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# Modern Control Technology: Components and Systems



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Kilian

# CHAPTER 1

## Introduction to Control Systems

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### OBJECTIVES

After studying this chapter, you should be able to:

- Distinguish between open-loop and closed-loop control systems.
- Understand control system block diagrams.
- Explain transfer functions.
- Differentiate between analog and digital control systems.
- Know how process control systems work.
- Know how servomechanisms work.

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### INTRODUCTION

A control system is a collection of components working together under the direction of some machine intelligence. In most cases, electronic circuits provide the intelligence, and electromechanical components such as sensors and motors provide the interface to the physical world. A good example is the modern automobile. Various sensors supply the on-board computer with information about the engine's condition. The computer then calculates the precise amount of fuel to be injected into the engine and adjusts the ignition timing. The mechanical parts of the system include the engine, transmission, wheels, and so on. To design, diagnose, or repair these sophisticated systems, you must understand the electronics, the mechanics, and control system principles.

In days past, so-called automatic machines or processes were controlled either by analog electronic circuits, or circuits using switches, relays, and timers. Since the advent of the inexpensive microprocessor, more and more devices and systems are being redesigned to incorporate a microprocessor controller. Examples include copying machines, soft-drink machines, robots, and industrial process controllers. Many of these machines are taking advantage of the increased processing power that comes with the microprocessor and, as a consequence, are becoming more sophisticated and are including new features. Taking again the modern automobile as an example, the original motivation for the on-board computer was to replace the mechanical and vacuum-driven

subsystems used in the distributor and carburetor. Once a computer was in the design, however, making the system more sophisticated was relatively easy—for example, self-adjusting fuel/air ratio for changes in altitude. Also, features such as computer-assisted engine diagnostics could be had without much additional cost. This trend toward computerized control will no doubt continue in the future.

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## 1.1 CONTROL SYSTEMS

### Introduction and Background

In a modern **control system**, electronic intelligence controls some physical process. Control systems are the “automatic” in such things as automatic pilot and automatic washer. Because the machine itself is making the routine decisions, the human operator is freed to do other things. In many cases, machine intelligence is better than direct human control because it can react faster or slower (keep track of long-term slow changes), respond more precisely, and maintain an accurate log of the system’s performance.

Control systems can be classified in several ways. A **regulator system** automatically maintains a parameter at (or near) a specified value. An example of this is a home-heating system maintaining a set temperature despite changing outside conditions. A **follow-up system** causes an output to follow a set path that has been specified in advance. An example is an industrial robot moving parts from place to place. An **event control system** controls a sequential series of events. An example is a washing machine cycling through a series of programmed steps.

Natural control systems have existed since the beginning of life. Consider how the human body regulates temperature. If the body needs to heat itself, food calories are converted to produce heat; on the other hand, evaporation causes cooling. Because evaporation is less effective (especially in humid climates), it is not surprising that our body temperature (98.6°F) was set near the high end of Earth’s temperature spectrum (to reduce demand on the cooling system). If temperature sensors in the body notice a drop in temperature, they signal the body to burn more fuel. If the sensors indicate too high a temperature, they signal the body to sweat.

Man-made control systems have existed in some form since the time of the ancient Greeks. One interesting device described in the literature is a pool of water that could never be emptied. The pool had a concealed float-ball and valve arrangement similar to a toilet tank mechanism. When the water level lowered, the float dropped and opened a valve that admitted more water.

Electrical control systems are a product of the twentieth century. Electromechanical relays were developed and used for remote control of motors and devices. Relays and switches were also used as simple logic gates to implement some intelligence. Using vacuum-tube technology, significant development in control systems was made during World War II. Dynamic position control systems (servomechanisms) were developed for aircraft applications, gun turrets, and torpedoes. Today, position control systems are

used in machine tools, industrial processes, robots, cars, and office machines, to name a few.

Meanwhile, other developments in electronics were having an impact on control system design. Solid-state devices started to replace the power relays in motor control circuits. Transistors and integrated circuit operational amplifiers (IC op-amps) became available for analog controllers. Digital integrated circuits replaced bulky relay logic. Finally, and perhaps most significantly, the microprocessor allowed for the creation of digital controllers that are inexpensive, reliable, able to control complex processes, and adaptable (if the job changes, the controller can be reprogrammed).

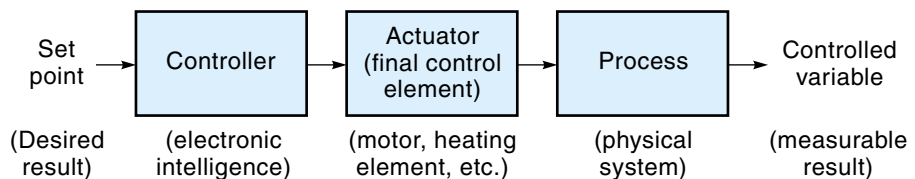
The subject of control systems is really many subjects: electronics (both analog and digital), power-control devices, sensors, motors, mechanics, and *control system theory*, which ties together all these concepts. Many students find the subject of control systems to be interesting because it deals with applications of much of the theory to which they have already been exposed. In this text, we will present material in each major subject area that makes up a control system, in more or less the same order that they appear in a control system block diagram. Some readers may choose to skip over (or lightly review) chapters that may be repetitious to them.

Finally, figures in this text use *conventional current flow*, current that travels from the positive to the negative terminal. If you are familiar with electron flow, remember that the theory and “numbers” are the same; only the indicated direction of the current is opposite from what you are used to.

Every control system has (at least) a **controller** and an **actuator** (also called a final control element). Shown in the block diagram in Figure 1.1, the controller is the intelligence of the system and is usually electronic. The input to the controller is called the **set point**, which is a signal representing the desired system output. The actuator is an electro-mechanical device that takes the signal from the controller and converts it into some kind of physical action. Examples of typical actuators would be an electric motor, an electrically controlled valve, or a heating element. The last block in Figure 1.1 is labeled **process** and has an output labeled **controlled variable**. The process block represents the physical process being affected by the actuator, and the controlled variable is the measurable result of that process. For example, if the actuator is an electric heating element in a furnace, then the process is “heating the furnace,” and the controlled variable is the temperature in the furnace. If the actuator is an electric motor that rotates an antenna, then the process is “rotating of the antenna,” and the controlled variable is the angular position of the antenna.

**Figure 1.1**

A block diagram of a control system.



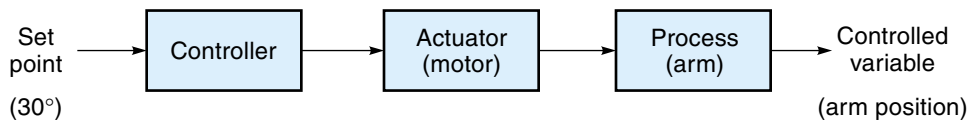
## Open-Loop Control Systems

Control systems can be broadly divided into two categories: open- and closed-loop systems. In an **open-loop control system**, the controller independently calculates exact voltage or current needed by the actuator to do the job and sends it. With this approach, however, *the controller never actually knows if the actuator did what it was supposed to* because there is no feedback. This system absolutely depends on the controller knowing the operating characteristics of the actuator.

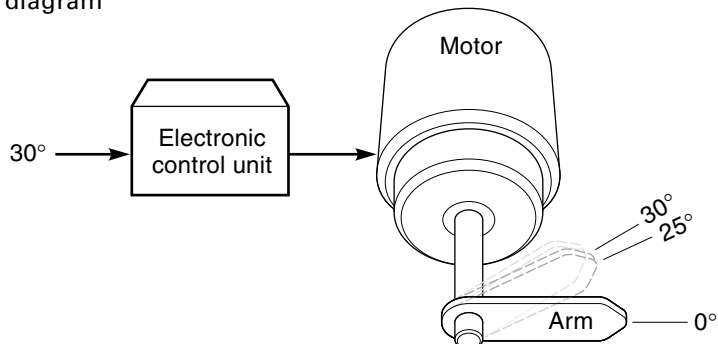
### EXAMPLE 1.1

Figure 1.2 shows an open-loop control system. The actuator is a motor driving a robot arm. In this case, the process is the arm moving, and the controlled variable is the angular position of the arm. Earlier tests have shown that the motor rotates the arm at 5 degrees/second (deg/s) at the rated voltage. Assume that the controller is directed to move the arm from  $0^\circ$  to  $30^\circ$ . Knowing the characteristics of the process, the controller sends a 6-second power pulse to the motor. If the motor is acting properly, it will rotate exactly  $30^\circ$  in the 6 seconds and stop. On particularly cold days, however, the lubricant is more viscous (thicker), causing more internal friction, and the motor rotates only  $25^\circ$  in the 6 seconds; the result is a  $5^\circ$  error. The controller has no way of knowing of the error and does nothing to correct it.

**Figure 1.2**  
Open-loop control system.



(a) Block diagram



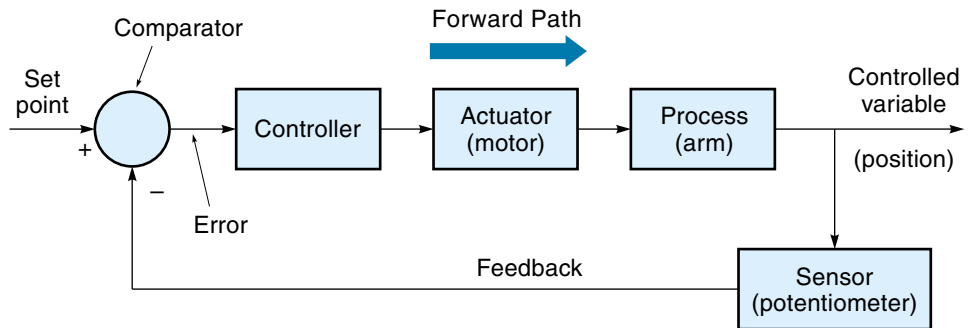
(b) A simple open-loop position system (Example 1.1)

Open-loop control systems are appropriate in applications where the actions of the actuator on the process are very repeatable and reliable. Relays and stepper motors (discussed in Chapters 4 and 8, respectively) are devices with reliable characteristics and are usually open-loop operations. Actuators such as motors or flow valves are sometimes used in open-loop operations, but they must be calibrated and adjusted at regular intervals to ensure proper system operation.

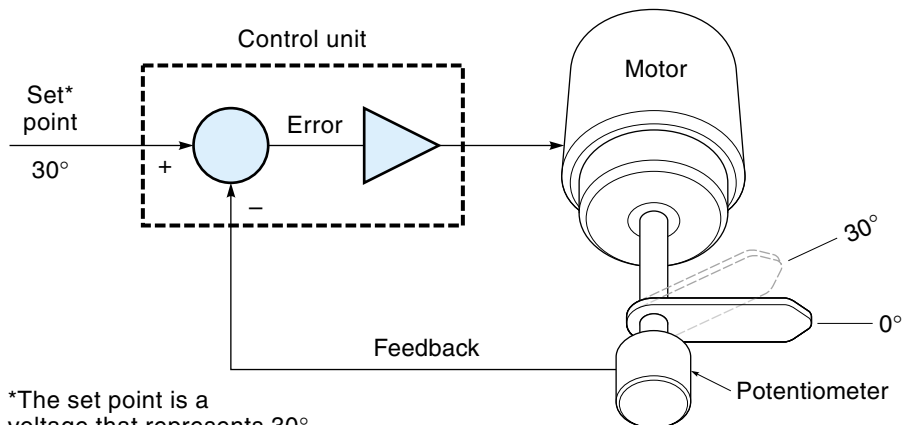
### Closed-Loop Control Systems

In a **closed-loop control system**, the output of the process (controlled variable) is constantly monitored by a **sensor**, as shown in Figure 1.3(a). The sensor samples the system output and converts this measurement into an electric signal that it passes back to the controller. Because the controller knows what the system is actually doing, it can make any adjustments necessary to keep the output where it belongs. The signal from the controller to the actuator is the **forward path**, and the signal from the sensor to the

**Figure 1.3**  
Closed-loop control system.



(a) Block diagram



\*The set point is a voltage that represents 30°

(b) A simple closed-loop position system (Example 1.2)

controller is the **feedback** (which “closes” the loop). In Figure 1.3(a), the feedback signal is subtracted from the set point at the **comparator** (just ahead of the controller). By subtracting the actual position (as reported by the sensor) from the desired position (as defined by the set point), we get the system **error**. The error signal represents the difference between “where you are” and “where you want to be.” *The controller is always working to minimize this error signal. A zero error means that the output is exactly what the set point says it should be.*

Using a **control strategy**, which can be simple or complex, the controller minimizes the error. A simple control strategy would enable the controller to turn the actuator on or off—for example, a thermostat cycling a furnace on and off to maintain a certain temperature. A more complex control strategy would let the controller adjust the actuator force to meet the demand of the load, as described in Example 1.2.

### EXAMPLE 1.2

As an example of a closed-loop control system, consider again the robot arm resting at  $0^\circ$  [see Figure 1.3(b)]. This time a potentiometer (pot) has been connected directly to the motor shaft. As the shaft turns, the pot resistance changes. The resistance is converted to voltage and then fed back to the controller.

To command the arm to  $30^\circ$ , a set-point voltage corresponding to  $30^\circ$  is sent to the controller. Because the actual arm is still resting at  $0^\circ$ , the error signal “jumps up” to  $30^\circ$ . Immediately, the controller starts to drive the motor in a direction to reduce the error. As the arm approaches  $30^\circ$ , the controller slows the motor; when the arm finally reaches  $30^\circ$ , the motor stops. If at some later time, an external force moves the arm off the  $30^\circ$  mark, the error signal would reappear, and the motor would again drive the arm to the  $30^\circ$  position.

The self-correcting feature of closed-loop control makes it preferable over open-loop control in many applications, despite the additional hardware required. This is because closed-loop systems provide reliable, repeatable performance even when the system components themselves (in the forward path) are not absolutely repeatable or precisely known.

## Transfer Functions

Physically, a control system is a collection of components and circuits connected together to perform a useful function. *Each component in the system converts energy from one form to another*; for example, we might think of a temperature sensor as converting degrees to volts or a motor as converting volts to revolutions per minute. To describe the performance of the entire control system, we must have some common language so that we can calculate the combined effects of the different components in the system. This need is behind the transfer function concept.



A **transfer function** (TF) is a mathematical relationship between the input and output of a control system component. Specifically, the transfer function is defined as the output divided by the input, expressed as

$$\text{TF} = \frac{\text{output}}{\text{input}} \quad (1.1)$$

Technically, the transfer function must describe both the time-dependent and the steady-state characteristics of a component. For example, a motor may have an initial surge of current that levels off at a lower steady-state value. The mathematics necessary to account for the time-dependent performance is beyond the scope of this text. In this text, we will consider only *steady-state values* for the transfer function, which is sometimes called simply the **gain**, expressed as

$$\text{TF}_{\text{steady-state}} = \text{gain} = \frac{\text{steady-state output}}{\text{steady-state input}} \quad (1.2)$$

### EXAMPLE 1.3

A potentiometer is used as a position sensor [see Figure 1.3(b)]. The pot is configured in such a way that  $0^\circ$  of rotation yields 0 V and  $300^\circ$  yields 10 V. Find the transfer function of the pot.

### SOLUTION

The transfer function is output divided by input. In this case, the input to the pot is “position in degrees,” and output is volts:

$$\text{TF} = \frac{\text{output}}{\text{input}} = \frac{10 \text{ V}}{300^\circ} = 0.0333 \text{ V/deg}$$

The transfer function of a component is an extremely useful number. It allows you to calculate the output of a component if you know the input. The procedure is simply to multiply the transfer function by the input, as shown in Example 1.4.

### EXAMPLE 1.4

For a temperature-measuring sensor, the input is temperature, and the output is voltage. The sensor transfer function is given as 0.01 V/deg. Find the sensor-output voltage if the temperature is  $600^\circ\text{F}$ .

### SOLUTION

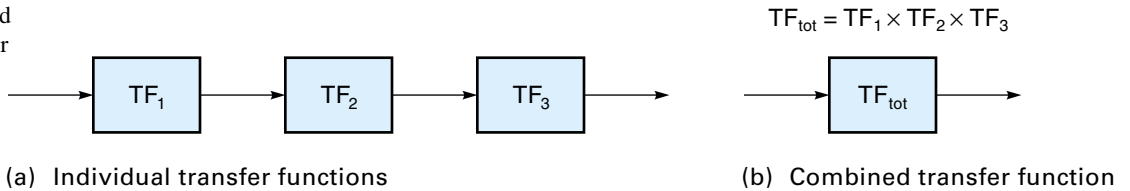
If  $\text{TF} = \frac{\text{output}}{\text{input}}$ , then

$$\text{Output} = \text{input} \times \text{TF}$$

$$= \frac{600^\circ \times 0.01 \text{ V}}{\text{deg}} = 6 \text{ V}$$

**Figure 1.4**

A series of transfer functions reduced to a single transfer function.



As mentioned previously, transfer functions can be used to analyze an entire system of components. One common situation involves a series of components where the output of one component becomes the input to the next and each component has its own transfer function. Figure 1.4(a) shows the block diagram for this situation. This diagram can be reduced into a single block that has a  $TF_{\text{tot}}$ , which is the product of all the individual transfer functions. This concept is illustrated in Figure 1.4(b) and stated in Equation 1.3:\*

$$TF_{\text{tot}} = \text{system gain} = TF_1 \times TF_2 \times TF_3 \times \dots \quad (1.3)$$

where

$TF_{\text{tot}}$  = total steady-state transfer function for the entire (open-loop) system

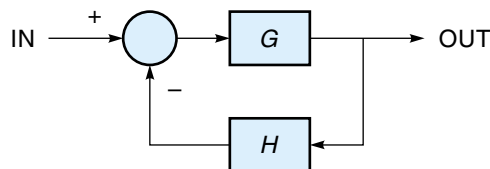
$TF_1, TF_2, \dots$  = individual transfer functions

These concepts are explained in Example 1.5.

### EXAMPLE 1.5

Consider the system shown in Figure 1.5. It consists of an electric motor driving a gear train, which is driving a winch. Each component has its own characteristics: The motor (under these conditions) turns at 100 rpm<sub>m</sub> for each volt ( $V_m$ ) supplied; the output shaft of the gear train rotates at one-half of the motor

\*Equation 1.3 is for open-loop systems only. If there is a feedback path (as shown in the accompanying diagram), then the overall system gain can be calculated as follows:  $TF_{\text{tot}} = G/(1 + GH)$ , where  $G$  is the total gain of the forward path and  $H$  is the total gain of the feedback path.



speed; the winch (with a 3-inch shaft circumference) converts the rotary motion ( $\text{rpm}_w$ ) to linear speed. The individual transfer functions are given as follows:

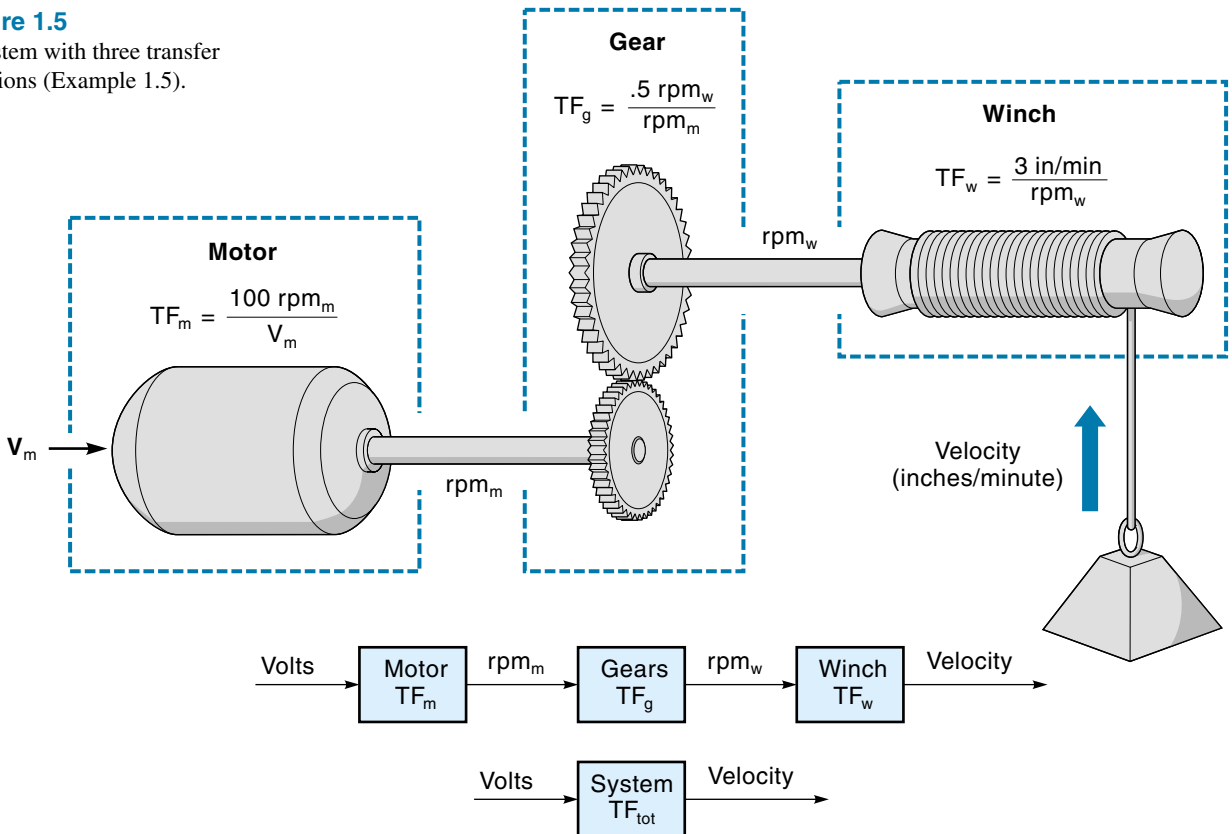
Motor: 
$$TF_m = \frac{\text{output}}{\text{input}} = \frac{100 \text{ rpm}_m}{1 \text{ V}_m} = 100 \text{ rpm}_m/\text{V}$$

Gear train: 
$$TF_g = \frac{\text{output}}{\text{input}} = \frac{1 \text{ rpm}_w}{2 \text{ rpm}_m} = 0.5 \text{ rpm}_w/\text{rpm}_m$$

Winch: 
$$TF_w = \frac{\text{output}}{\text{input}} = \frac{3 \text{ in./min}}{1 \text{ rpm}_w} = 3 \text{ in./min}/\text{rpm}_w$$

Using Equation 1.3, we can calculate the system transfer function. If everything is correct, all units will cancel except for the desired set:

**Figure 1.5**  
A system with three transfer functions (Example 1.5).



$$\begin{aligned}
 TF_{tot} &= TF_m \times TF_g \times TF_w \\
 &= \frac{100 \text{ rpm}_m}{1 \text{ V}_m} \times \frac{0.5 \text{ rpm}_w}{1 \text{ rpm}_m} \times \frac{3 \text{ in./min}}{1 \text{ rpm}_w} \\
 &= 150 \text{ in./min/V}_m
 \end{aligned}$$

We have shown that the transfer function of the complete system is 150 in./min/V<sub>m</sub>. Knowing this value, we can calculate the system output for any system input. For example, if the input to the this system is 12 V (to the motor), the output speed of the winch is calculated as follows:

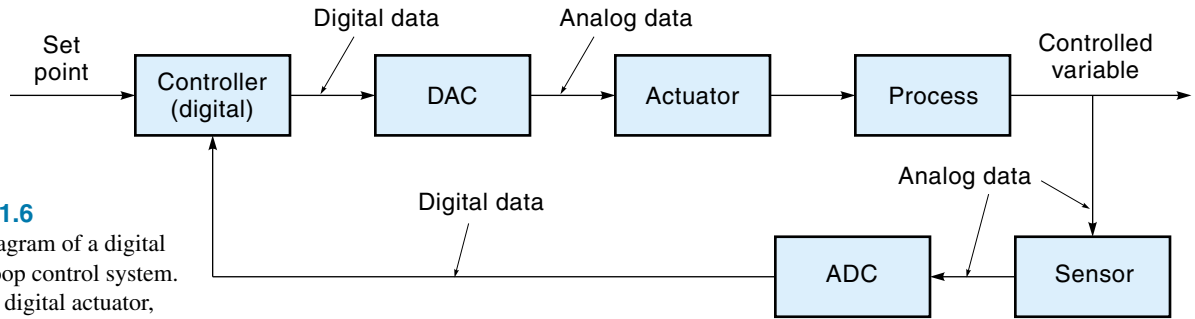
$$\text{Output} = \text{input} \times TF = \frac{12 \text{ V} \times 150 \text{ in./min}}{1 \text{ V}_m} = 1800 \text{ in./min}$$

## 1.2 ANALOG AND DIGITAL CONTROL SYSTEMS

In an **analog control system**, the controller consists of traditional analog devices and circuits, that is, linear amplifiers. The first control systems were analog because it was the only available technology. In the analog control system, any change in either set point or feedback is sensed immediately, and the amplifiers adjust their output (to the actuator) accordingly.

In a **digital control system**, the controller uses a digital circuit. In most cases, this circuit is actually a computer, usually microprocessor- or microcontroller-based. The computer executes a program that repeats over-and-over (each repetition is called an **iteration** or **scan**). The program instructs the computer to read the set point and sensor data and then use these numbers to calculate the controller output (which is sent to the actuator). The program then loops back to the beginning and starts over again. The total time for one pass through the program may be less than 1 millisecond (ms). The digital system only “looks” at the inputs at a certain time in the scan and gives the updated output later. If an input changes just after the computer looked at it, that change will remain undetected until the next time through the scan. This is fundamentally different than the analog system, which is continuous and responds immediately to any changes. However, for most digital control systems, the scan time is so short compared with the response time of the process being controlled that, for all practical purposes, the controller response is instantaneous.

The physical world is basically an “analog place.” Natural events take time to happen, and they usually move in a continuous fashion from one position to the next. Therefore, most control systems are controlling analog processes. This means that, in many cases, the digital control system must first convert real-world analog input data into digital form before it can be used. Similarly, the output from the digital controller must be converted from digital form back into analog form. Figure 1.6 shows a block diagram of a digital closed-loop control system. Notice the two additional blocks: the digital-to-analog converter (DAC) and the analog-to-digital converter (ADC). (These



**Figure 1.6**  
Block diagram of a digital closed-loop control system. (Note: A digital actuator, such as a stepper motor, would not need a DAC; similarly, a digital sensor, such as an optical shaft encoder, would not need an ADC.)

devices, which convert data between the digital and analog formats, are discussed in Chapter 2.) Also note that the feedback line is shown going directly into the controller. This emphasizes the fact that the computer, not a separate subtraction circuit, makes the comparison between the set point and the feedback signal.

### 1.3 CLASSIFICATIONS OF CONTROL SYSTEMS

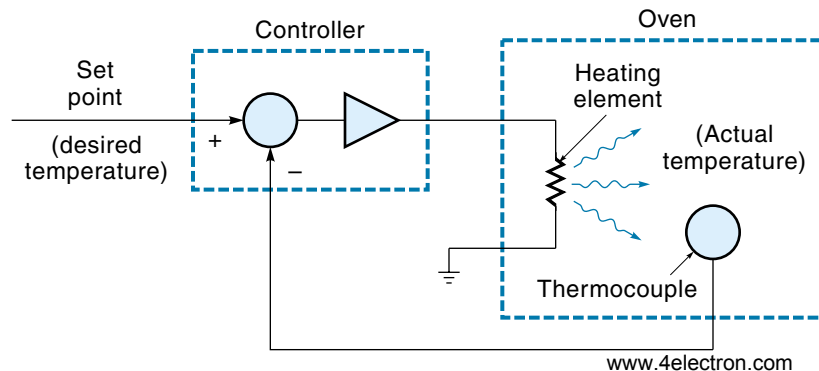
So far we have discussed control systems as being either open or closed loop, analog or digital. Yet we can classify control systems in other ways, which have to do with applications. Some of the most common applications are discussed next.

#### Process Control

**Process control** refers to a control system that oversees some industrial process so that a uniform, correct output is maintained. It does this by monitoring and adjusting the control parameters (such as temperature or flow rate) to ensure that the output product remains as it should.

The classic example of process control is a closed-loop system maintaining a specified temperature in an electric oven, as illustrated in Figure 1.7. In this case, the

**Figure 1.7**  
A closed-loop oven-heating system.



actuator is the heating element, the controlled variable is the temperature, and the sensor is a thermocouple (a device that converts temperature into voltage). The controller regulates power to the heating element in such a way as to keep the temperature (as reported by the thermocouple) at the value specified by the set point.

Another example of process control is a paint factory in which two colors, blue and yellow, are mixed to produce green (Figure 1.8). To keep the output color constant, the exact proportions of blue and yellow must be maintained. The setup illustrated in Figure 1.8(a) accomplishes this with flow valves 1 and 2, which are manually adjusted until the desired hue of green is achieved. The problem is that, as the level of paint in the vats changes, the flow will change and the mixture will not remain constant.

To maintain an even flow from the vats, we could add two electrically operated flow valves (and their controls) as shown in Figure 1.8(b). Each valve would maintain a specified flow of paint into the mixer, regardless of the upstream pressure. Theoretically, if the blue and yellow flows are independently maintained, the green should stay constant. In practice, however, other factors such as temperature or humidity may affect the mixing chemistry and therefore the output color.

A better approach might be the system shown in Figure 1.8(c); a single sensor monitors the output color. If the green darkens, the controller increases the flow of yellow. If the green gets too light, the flow of yellow is decreased. This system is desirable because it monitors the actual parameter that needs to be maintained. In real life, such a straightforward system may not be possible because sensors that can measure the output directly may not exist and/or the process may involve many variables.

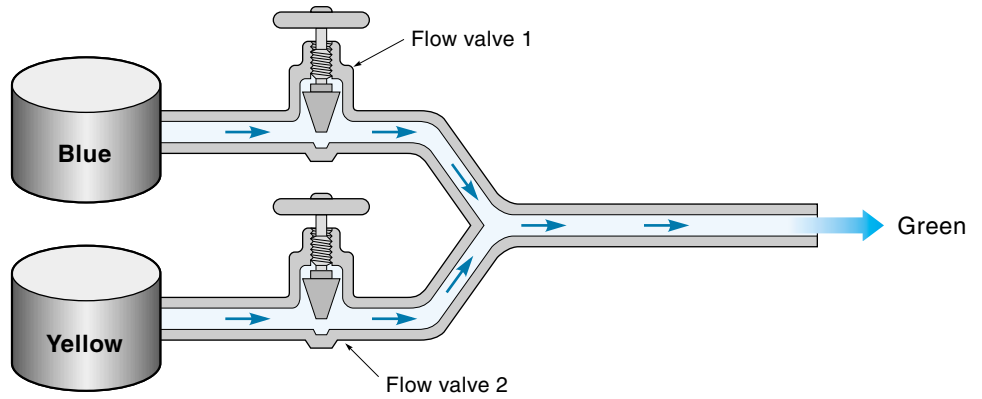
Process control can be classified as being a batch process or a continuous process. In a **continuous process** there is a continuous flow of material or product, as in the paint-mixing example just described. A **batch process** has a beginning and an end (which is usually performed over and over). Examples of batch processes include mixing a batch of bread dough and loading boxes on a pallet.

In a large plant such as a refinery, many processes are occurring simultaneously and must be coordinated because the output of one process is the input of another. In the early days of process control, separate independent controllers were used for each process, as shown in Figure 1.9(a). The problem with this approach was that, to change the overall flow of the product, each controller had to be readjusted manually.

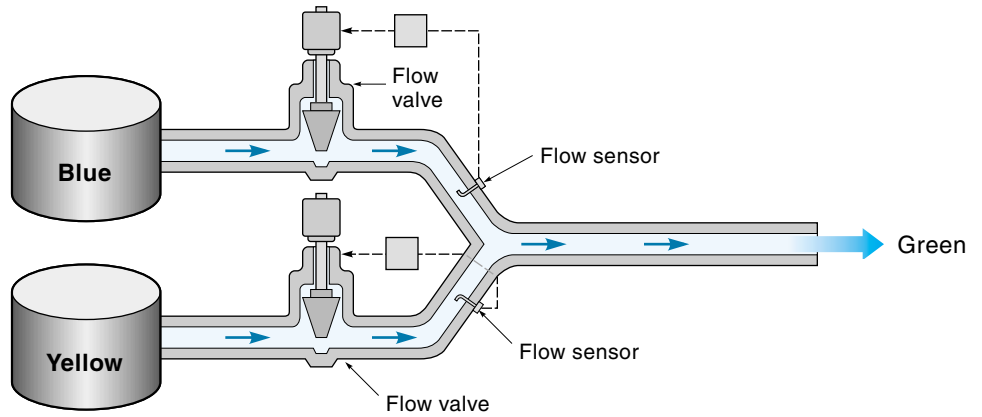
In the 1960s, a new system was developed in which all independent controllers were replaced by a single large computer. Illustrated in Figure 1.9(b), this system is called **direct digital control (DDC)**. The advantage of this approach is that all local processes can be implemented, monitored, and adjusted from the same place. Also, because the computer can “see” the whole system, it is in a position to make adjustments to enhance total system performance. The drawback is that the whole plant is dependent on that one computer. If the computer goes off line to fix a problem in one process, the whole plant shuts down.

**Figure 1.8**

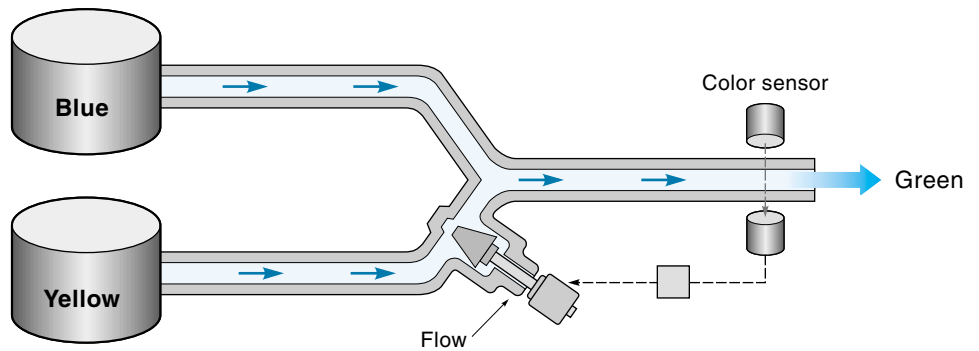
Process control in mixing paint.



(a) Manual control



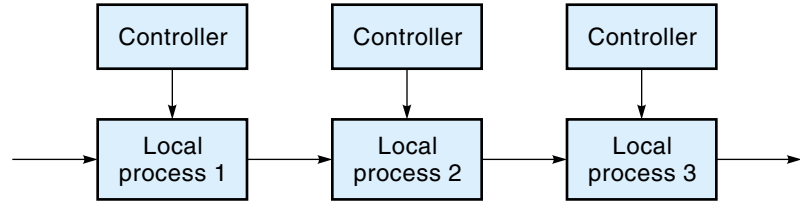
(b) Automatic flow control



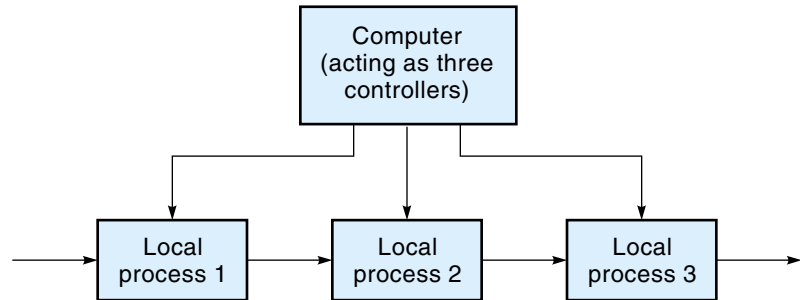
(c) Automatic color control

**Figure 1.9**

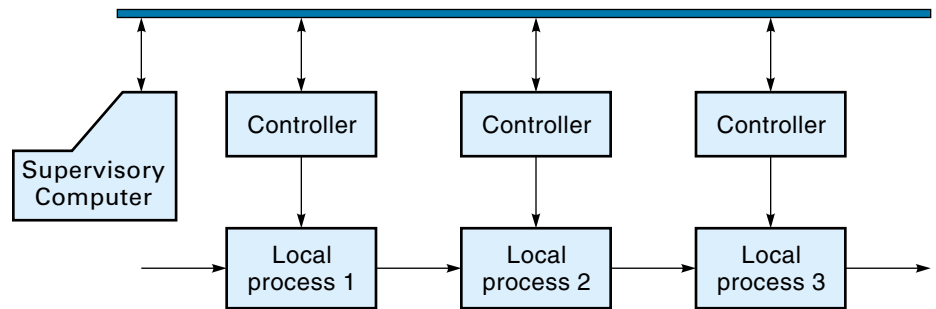
Approaches of multiprocess control.



(a) Individual local controllers



(b) Direct computer control of three processes



(c) Distributed computer control using local controllers

The advent of small microprocessor-based controllers has led to a new approach called **distributed computer control (DCC)**, illustrated in Figure 1.9(c). In this system, each process has its own separate controller located at the site. These local controllers are interconnected via a local area network so that all controllers on the network can be monitored or reprogrammed from a single supervisory computer. Once programmed, each process is essentially operating independently.

This makes for a more robust and safe system, because all the local processes will continue to function even if the supervisory computer or network goes down. For example, a local controller whose job it is to keep some material at a critical



temperature will continue to function even if the supervisory computer is temporarily disabled.

Increasingly, the components of a control system are being interconnected with the “business office” network in a factory, which allows the status of any process in the factory to be examined by any computer on anyone’s desk. You might be able to sit down at a PC anywhere in the building and determine whether a particular photo sensor on an assembly line has a dirty lens or how much current a particular motor is drawing.

**Sequentially Controlled Systems**

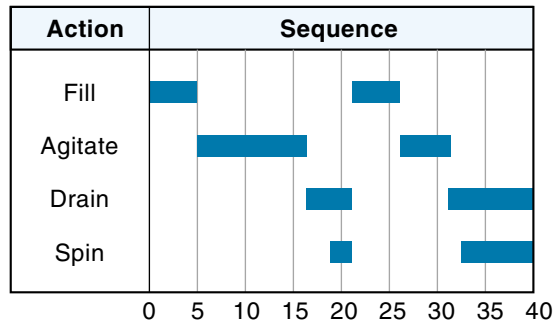
A **sequentially controlled system** controls a process that is defined as a series of tasks to be performed—that is, a sequence of operations, one after the other. Each operation in the sequence is performed either for a certain amount of time, in which case it is **time-driven**, or until the task is finished (as indicated by, say, a limit switch), in which case it is **event-driven**. A time-driven sequence is *open-loop* because there is no feedback, whereas an event-driven task is *closed-loop* because a feedback signal is required to specify when the task is finished.

The classic example of a sequentially controlled system is the automatic washing machine. The first event in the wash cycle is to fill the tub. This is an event-driven task because the water is admitted until it gets to the proper level as indicated by a float and limit switch (closed loop). The next two tasks, wash and spin-drain, are each done for a specified period of time and are time-driven events (open loop). A timing diagram for a washing machine is shown in Figure 1.10.

Another example of a sequentially controlled system is a traffic signal. The basic sequence may be time-driven: 45 seconds for green, 3 seconds for yellow, and 45 seconds for red. The presence or absence of traffic, as indicated by sensors in the roadbed, however, may alter the basic sequence, which is an event-driven control.

Many automated industrial processes could be classified as sequentially controlled systems. An example is a process where parts are loaded into trays, inserted into a furnace for 10 minutes, then removed and cooled for 10 minutes, and loaded into boxes in groups of six. In the past, most sequentially controlled systems used switches, relays, and electromechanical timers to implement the control logic. These tasks are now performed

**Figure 1.10**  
Timing diagram for an automatic washing machine.



more and more by small computers known as **programmable logic controllers** (PLCs), which are less expensive, more reliable, and easily reprogrammed to meet changing needs—for example, to put eight items in a box instead of six. (PLCs are discussed in Chapter 12.)

## Motion Control

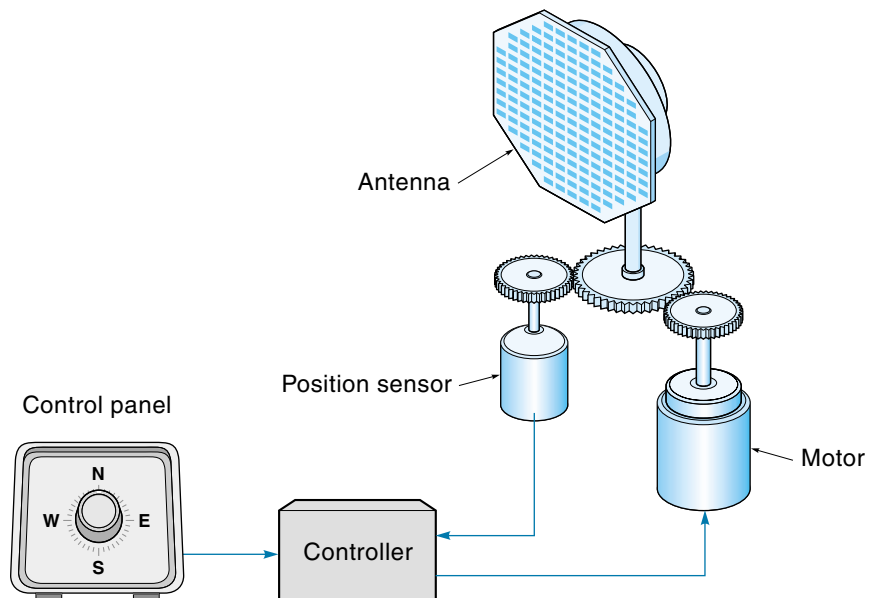
**Motion control** is a broad term used to describe an open-loop or closed-loop electro-mechanical system wherein things are moving. Such a system typically includes a motor, mechanical parts that move, and (in many cases) feedback sensor(s). Automatic assembling machines, industrial robots, and numerical control machines are examples.

### *Servomechanisms*

**Servomechanism** is the traditional term applied to describe a closed-loop electro-mechanical control system that directs the precise movement of a physical object such as a radar antenna or robot arm. Typically, either the output position or the output velocity (or both) is controlled. An example of a servomechanism is the positioning system for a radar antenna, as shown in Figure 1.11. In this case, the controlled variable is the antenna position. The antenna is rotated with an electric motor connected to the controller located some distance away. The user selects a direction, and the controller directs the antenna to rotate to a specific position.

**Figure 1.11**

A servomechanism:  
a remote antenna-  
positioning system.

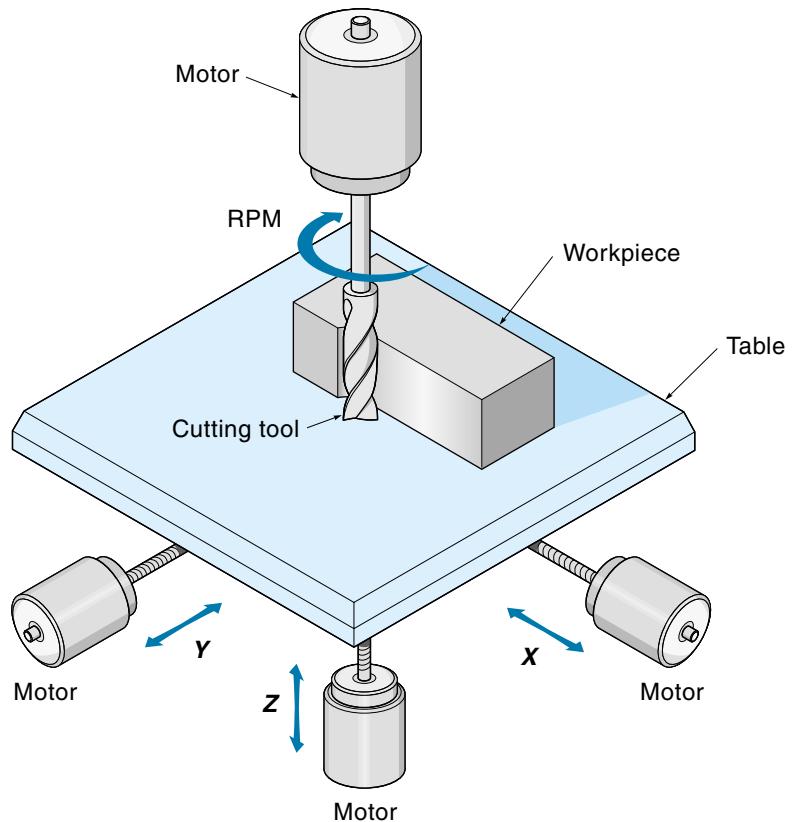


### Numerical Control

**Numerical control (NC)** is the type of digital control used on machine tools such as lathes and milling machines. These machines can automatically cut and shape the workpiece without a human operator. Each machine has its own set of axes or parameters that must be controlled; as an example, consider the milling machine shown in Figure 1.12. The workpiece that is being formed is fastened to a movable table. The table can be moved (with electric motors) in three directions:  $X$ ,  $Y$ , and  $Z$ . The cutting-tool speed is automatically controlled as well. To make a part, the table moves the workpiece past the cutting tool at a specified velocity and cutting depth. In this example, four parameters ( $X$ ,  $Y$ ,  $Z$ , and rpm) are continuously and independently controlled by the controller. The controller takes as its input a series of numbers that completely describe how the part is to be made. These numbers include the physical dimensions and such details as cutting speeds and feed rates.

NC machines have been used since the 1960s, and certain standards that are unique to this application have evolved. Traditionally, data from the part drawing were

**Figure 1.12**  
Basics of a numerical-control milling machine.



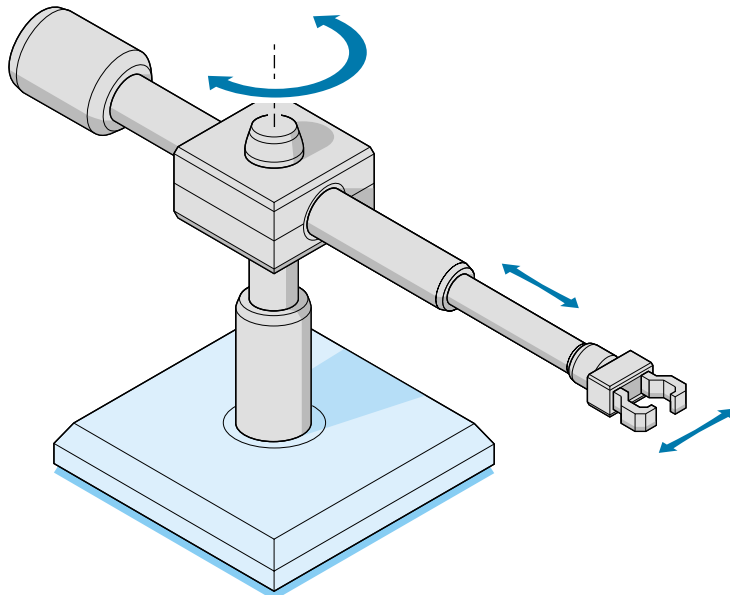
entered manually into a computer program. This program converted the input data into a series of numbers and instructions that the NC controller could understand, and either stored them on a floppy disk or tape, or sent the data directly to the machine tool. These data were read by the machine-tool controller as the part was being made. With the advent of **computer-aided design (CAD)**, the job of manually programming the manufacturing instructions has been eliminated. Now it is possible for a special computer program (called a *postprocessor*) to read the CAD-generated drawing and then produce the necessary instructions for the NC machine to make the part. This whole process—from CAD to finished part—is called **computer-aided manufacturing (CAM)**.

One big advantage of this process is that one machine tool can efficiently make many different parts, one after the other. This system tends to reduce the need for a large parts inventory. If the input tape (or software) is available, any needed part can be made in a short period of time. This is one example of **computer-integrated manufacturing (CIM)**, a whole new way of doing things in the manufacturing industry. CIM involves using the computer in every step of the manufacturing operation—from the customer order, to ordering the raw materials, to machining the part, to routing it to its final destination.

### Robotics

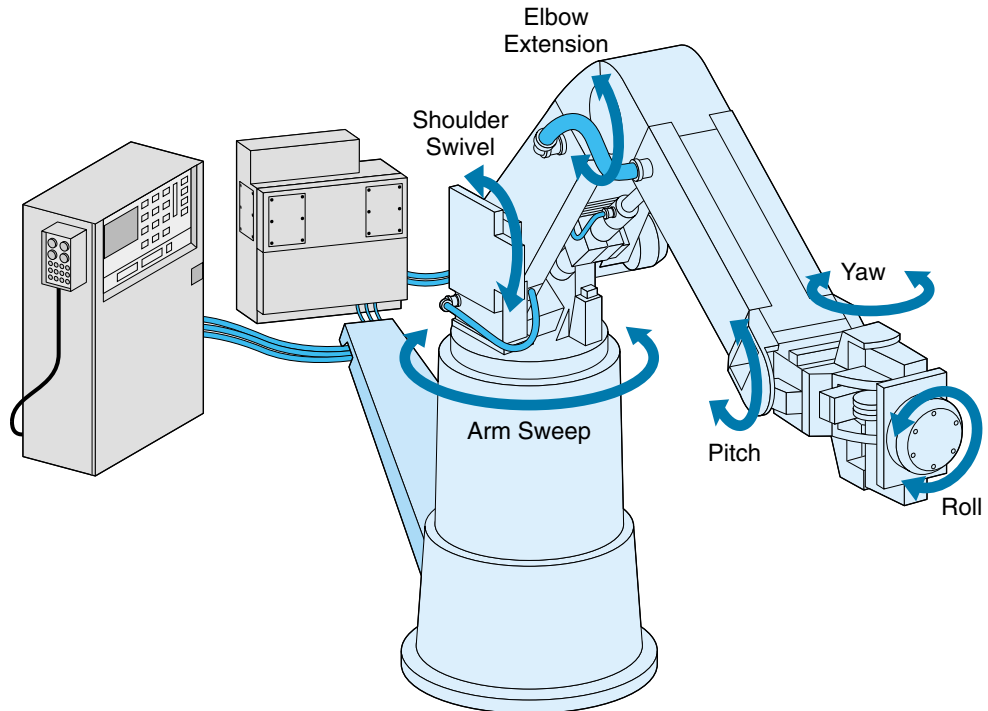
Industrial **robots** are classic examples of position control systems. In most cases, the robot has a single arm with shoulder, elbow, and wrist joints, as well as some kind of hand known as an *end effector*. The end effector is either a gripper or other tool such

**Figure 1.13**  
A pick-and-place robot.



**Figure 1.14**

A large industrial robot.



as a paint spray gun. Robots are used to move parts from place to place, assemble parts, load and off-load NC machines, and perform such tasks as spray painting and welding.

**Pick-and-place robots**, the simplest type, pick up parts and place them somewhere else nearby. Instead of using sophisticated feedback control, they are often run open-loop using mechanical stops or limit switches (discussed in Chapter 4) to determine how far in each direction to go (sometimes called a “bang-bang” system). An example is shown in Figure 1.13. This robot uses pneumatic cylinders to lift, rotate, and extend the arm. It can be programmed to repeat a simple sequence of operations.

Sophisticated robots use closed-loop position systems for all joints. An example is the industrial robot shown in Figure 1.14. It has six independently controlled axes (known as six degrees of freedom) allowing it to get to difficult-to-reach places. The robot comes with and is controlled by a dedicated computer-based controller. This unit is also capable of translating human instructions into the robot program during the “teaching” phase. The arm can move from point to point at a specified velocity and arrive within a few thousandths of an inch.

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## SUMMARY

A control system is a system where electronic intelligence controls some physical process. This text will deal with all phases of the control system: the electronics, the

power sources (such as motors), the mechanics, and control system theory, which ties together all the concepts.

A control system is described in terms of a block diagram. The first block is the controller, which represents the electronic intelligence. The controller outputs a control signal to the next block, which is the actuator. The actuator is the system's first physical device to do something (for example, a motor or heating element).

There are two general categories of control systems: open loop and closed loop. In open-loop control, the controller sends a measured signal, specifying the desired action, to the actuator (the controller, however, has no way of knowing what the actuator actually does). Closed-loop control includes a sensor that feeds back a signal from the actuator to the controller, informing the controller exactly what the output is doing. This allows the controller to make correctional adjustments.

Each component in the control system can be described mathematically by a transfer function (TF), where  $TF = \text{output}/\text{input}$ . Transfer functions of individual components in a system can be mathematically combined to calculate overall system performance. A true transfer function includes time-dependent and steady-state characteristics, whereas a useful simplification (employed in this text) considers only steady-state conditions.

Control systems are classified as analog or digital. In an analog control system, the controller uses traditional analog electronic circuits such as linear amplifiers. In a digital control system, the controller uses a digital circuit, usually a computer.

Control systems are classified by application. Process control usually refers to an industrial process being electronically controlled for the purpose of maintaining a uniform correct output. Motion control refers to a system wherein things move. A servo-mechanism is a feedback control system that provides remote control motion of some object, such as a robot arm or a radar antenna. A numerical control (NC) control system directs a machine tool, such as a lathe, to machine a part automatically.

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## GLOSSARY

**actuator** The first component in the control system which generates physical movement, typically a motor. The actuator gets its instructions directly from the controller. Another name for the actuator is the *final control element*

**analog control system** A control system where the controller is based on analog electronic circuits, that is, linear amplifiers.

**batch process** A process that has a beginning and an end and is usually preformed over and over.

**CAD** See **computer-aided design**.

**CAM** See **computer-aided manufacturing**.

**CIM** See **computer-integrated manufacturing**.

**closed-loop control system** A control system that uses feedback. A sensor continually monitors the output of the system and sends a signal to the controller, which makes adjustments to keep the output within specification.

**comparator** Part of the control system that subtracts the feedback signal (as reported by the sensor) from the set point, to determine the error.

**computer-aided design** A computer system that makes engineering drawings.

**computer-aided manufacturing** A computer system that allows CAD drawings to be converted for use by a numerical control (NC) machine tool.

**computer-integrated manufacturing** A computer system that oversees every step in the manufacturing process, from customer order to delivery of finished parts.

**continuous process** A process wherein there is a continuous flow of product—for example, a steam boiler where water is continuously pumped in and steam continuously comes out.

**controlled variable** The ultimate output of the process; the actual parameter of the process that is being controlled.

**controller** The machine intelligence of the control system.

**control strategy** The set of rules that the controller follows to determine its output to the actuator.

**control system** A system that may include electronic and mechanical components, where some type of machine intelligence controls a physical process.

**DCC** See **distributed computer control**.

**DDC** See **direct digital control**.

**digital control system** A control system where the controller is a digital circuit, typically a computer.

**direct digital control** An approach to process control where all controllers in a large process are simulated by a single computer.

**distributed computer control** An approach to process control where each process has its own local controller, but all individual controllers are connected to a single computer for programming and monitoring.

**error** In a control system, the difference between where the system is supposed to be (set point) and where it really is.

**event control system** A control system that cycles through a predetermined series of steps.

**event-driven operation** In a sequentially controlled system, an action that is allowed to start or continue based on some parameter changes. This is an example of closed-loop control.

**feedback** The signal from the sensor, which is fed back to the controller.

**follow-up system** A control system where the output follows a specified path.

**forward path** The signal-flow direction of the controller to the actuator.

**gain** The steady-state relationship between input and output of a component. (In this text, *gain* and *transfer function* are used interchangeably, although this is a simplification.)

**iteration** See **scan**.

**motion control** A term that refers to an electromechanical system wherein things move.

**NC** See **numerical control**.

**numerical control** A digital control system that directs machine tools, such as a lathe, to automatically machine a part.

**open-loop control system** A control system that does not use feedback. The controller sends a measured signal to the actuator, which specifies the desired action. This type of system is *not* self-correcting.

**pick-and-place robot** A simple robot that does a repetitive task of picking up and placing an object somewhere else.

**PLC** See **programmable logic controller**.

**programmable logic controller** A small, self-contained microprocessor-based controller used primarily to replace relay logic controllers.

**process** The physical process that is being controlled.

**process control** A control system that maintains a uniform, correct output for some industrial process.

**regulator system** A control system that maintains an output at a constant value.

**robot** A servomechanism type control system in the form of a machine with a movable arm.

**scan** One cycle through the program loop of a computer-based controller.

**sensor** Part of the control system that monitors the system output, the sensor converts the physical output action of the system into an electric signal, which is fed back to the controller.

**sequentially controlled system** A control system that performs a series of actions in sequence, an example being a washing machine.



**servomechanism** An electromechanical feedback control system where the output is linear or rotational movement of a mechanical part.

**set point** The input signal to the control system, specifying the desired system output.

**time-driven operation** In a sequentially controlled system, an action that is allowed to happen for a specified period of time. This is an example of open-loop control.

**TF** See **transfer function**.

**transfer function** A mathematical relationship between the input and output of a control system component:  $TF = \text{output}/\text{input}$ . (In this text, *transfer function* and *gain* are used interchangeably, although this is a simplification.)

## EXERCISES

### Section 1.1

1.
  - a. Draw a block diagram of an open-loop control system.
  - b. Use the block diagram to describe how the system works.
  - c. What basic requirements must the components meet for this system to work?
  - d. What is the advantage of this system over a closed-loop system?
2.
  - a. Draw a block diagram of a closed-loop control system.
  - b. Use the block diagram to describe how the system works.
  - c. What is the advantage of this system over an open-loop system?
3. The controlled variable in a closed-loop system is a robot arm. Initially, it is at  $45^\circ$ ; then it is commanded to go to  $30^\circ$ . Describe what happens in terms of set point, feedback signal, error signal, and arm position.
4. Identify the following as open- or closed-loop control.
  - a. Controlling the water height in a toilet tank
  - b. Actuation of street lights at 6 P.M.
  - c. Stopping a clothes dryer when the clothes are dry
  - d. Actuation of an ice maker when the supply of cubes is low
5. A potentiometer has a transfer function of  $0.1 \text{ V/deg}$ . Find the pot's output if the input is  $45^\circ$ .
6. A potentiometer has a transfer function of  $0.05 \text{ V/deg}$ . Find the pot's output if the input is  $89^\circ$ .
7. A motor was measured to rotate (unloaded) at 500 rpm with a 6-V input and 1000 rpm with a 12-V input. What is the transfer function (steady state) for the unloaded motor?
8. In a certain system, an electric heating element was found to increase the temperature of a piece of metal  $10^\circ$  for each ampere of current. The metal expands 0.001 in./deg and pushes on a load sensor which outputs  $1 \text{ V}/0.005 \text{ in.}$  of compression.

- a. Find the transfer functions of the three components and draw the block diagram.
- b. Calculate the overall transfer function of this system.

### Section 1.2

9. Describe the differences between an *analog* and a *digital control system*
10. The iteration time of a digital controller is 1 s. Would this controller be appropriate for the following?
  - a. A robot that paint sprays cars
  - b. A solar panel control system that tracks the sun across the sky

### Section 1.3

11. What is the difference between a *process control system* and a *servomechanism*
12. What is the difference between *direct digital control* and *distributed computer control*
13. Give an example (other than in this book) of the following:
  - a. A time-driven control system
  - b. An event-driven control system
  - c. A combined time- and event-driven control system
14. Give an example (other than in this book) of a servomechanism.

# CHAPTER 2

## Introduction to Microprocessor-Based Control

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### OBJECTIVES

After studying this chapter, you should be able to:

- Understand what a microprocessor is, what it does, and how it works.
- Understand the concepts of RAM and ROM computer memory and how memory is accessed via the address and data buses.
- Understand how parallel and serial data interfaces work.
- Perform relevant calculations pertaining to analog-to-digital converters and digital-to-analog converters.
- Understand the principles of digital controller software.
- Recognize and describe the characteristics of the various types of available digital controllers, that is, microcontrollers, single-board computers, programmable logic controllers, and personal computers.

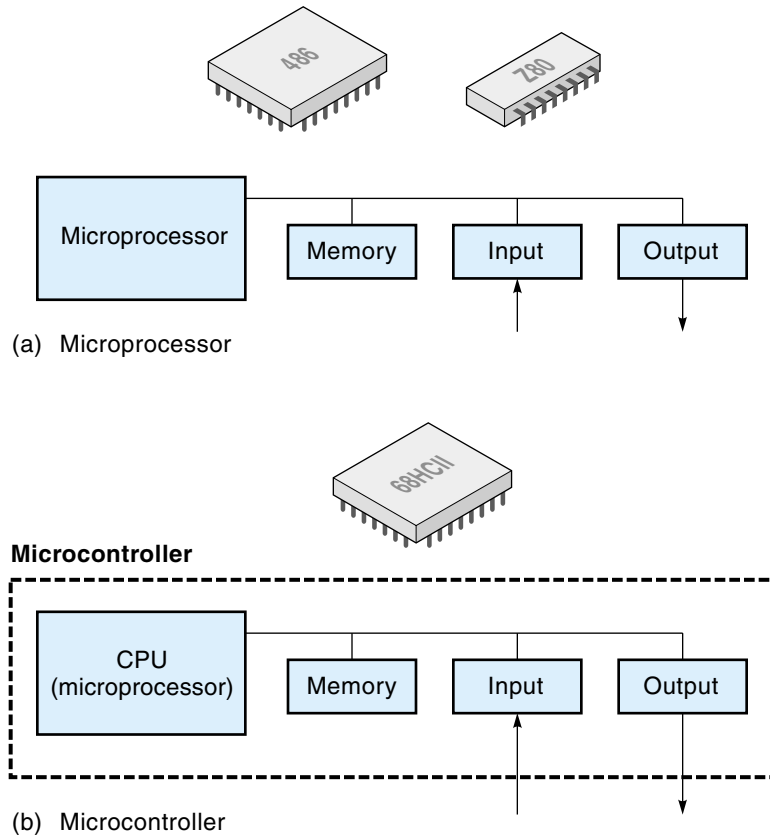
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### INTRODUCTION

The digital integrated circuit (IC) called a **microprocessor** [Figure 2.1(a)], has ushered in a whole new era for control systems electronics. This revolution has occurred because the microprocessor brings the flexibility of program control and the computational power of a computer to bear on any problem. Automatic control applications are particularly well suited to take advantage of this technology, and microprocessor-based control systems are rapidly replacing many older control systems based on analog circuits or electromechanical relays. One of the first microprocessor-based controllers made specifically for control applications was the programmable logic controller (PLC), which is discussed later in this chapter and in Chapter 12. A microprocessor by itself is not a computer; additional components such as memory and input/output circuits are required to make it operational. However, the **microcontroller** [Figure 2.1(b)], which is

**Figure 2.1**

Microprocessor and microcontroller.



a close relative of the microprocessor, *does* contain all the computer functions on a single IC. Microcontrollers lack some of the power and speed of the newer microprocessors, but their compactness is ideal for many control applications; most so-called microprocessor-controlled devices, such as vending machines, are really using microcontrollers. Some specific reasons for using a digital, microprocessor design in control systems are the following:

- Low-level signals from sensors, once converted to digital, can be transmitted long distances virtually error-free.
- A microprocessor can easily handle complex calculations and control strategies.
- Long-term memory is available to keep track of parameters in slow-moving systems.
- Changing the control strategy is easy by loading in a new program; no hardware changes are required.
- Microprocessor-based controllers are more easily connected to the computer network within an organization. This allows designers to enter program changes and read current system status from their desk terminals.

In this chapter, we will present the basic concepts of a microprocessor- and microcontroller-based system with particular emphasis on control system applications. It is by no means an in-depth treatment, but enough to make the rest of the text more meaningful.

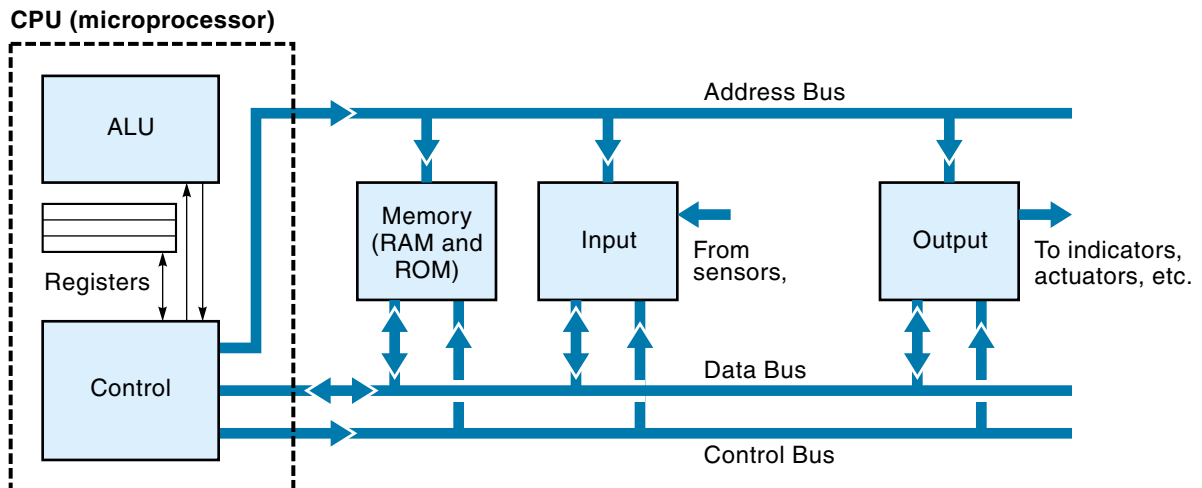
In the first sections of this chapter the basic concepts of microprocessor hardware and operation are introduced (these concepts also apply to microcontrollers). I have included this material because the student of modern control systems should have at least a general knowledge of how the microprocessor performs its job.

## 2.1 INTRODUCTION TO MICROPROCESSOR SYSTEM HARDWARE

A computer is made up of four basic functional units: the central processing unit (CPU), memory, input, and output (I/O). The **central processing unit** does the actual computing and is composed of two subparts: the arithmetic logic unit and control sections (Figure 2.2). The arithmetic logic unit (ALU) performs the actual numerical and logic calculations such as addition, subtraction, AND, OR, and so on. The control section of the CPU manages the data flow, such as reading and executing the program instructions. If data require calculations, the control section hands it over to the ALU for processing. In a microprocessor-based computer, the microprocessor is the CPU.

**Figure 2.2**

A block diagram of a microprocessor-based computer.



Digital data is in the form of **bits**, where each bit has a value of either 1 or 0. Digital circuits usually use 5 Vdc to represent logic 1 and 0 Vdc to represent logic 0. Eight bits together is called a **byte**. A microprocessor handles digital data in **words**, where a word may be 8, 16, or 32 bits wide. For example, an 8-bit microprocessor has a byte-sized word, with a maximum decimal value of 255. (Computers represent numbers in the *binary number system*; for example, 11111111 binary = 255 decimal.) The rightmost bit in a binary number has the least value (usually 1) and is called the **least significant bit (LSB)**. The leftmost bit represents the highest value and is called the **most significant bit (MSB)**. The conversion between binary and decimal can be performed directly with most scientific calculators or manually using the technique shown in Example 2.1. To express values larger than 255, two or more words are put together. In this text, we will assume 8-bit microprocessors are used unless otherwise stated.

### EXAMPLE 2.1

Find the decimal value of the 8-bit binary number 10110011.

### SOLUTION

Each bit in the binary number has a different value, or weight. The LSB has a weight of 1. The bit to the left of the LSB has a weight of 2, the third bit has a weight of 4, and so on, with the weight doubling for each bit up to 128 for the MSB. To find the value of an 8-bit number, you can set up a chart (shown below) and then sum the values that correspond to the 1s in the binary number.

MSB							LSB	
128	64	32	16	8	4	2	1	← Bit weights
1	0	1	1	0	0	1	1	← Binary number
								1 × 1 = 1
								2 × 1 = 2
								4 × 0 = 0
								8 × 0 = 0
								16 × 1 = 16
								32 × 1 = 32
								64 × 0 = 0
								128 × 1 = 128
								+
								179
								decimal number

The **memory** section of the computer is a place where digital data in binary form (1s and 0s) are stored. Memory consists of cells organized in 8-bit groups. Each byte

is given a unique numeric **address**, which represents its location just as a street address represents the location of a house. Data are written into memory and read out of memory, based solely on their address. In a particular memory circuit, the addresses might start at 1000 and run consecutively to 2000. Figure 2.3 diagrams a section of memory. Note that the first byte of data has a decimal value of 2 ( $00000010 = 2$  decimal) and an address of 1000.

Computers usually have two kinds of addressable memory. The first is **random-access memory** (RAM), which allows the computer to read and write data at any of its addresses (it is also called read/write memory, or RWM). All data in this type of memory are lost when the power is turned off and is called **volatile memory** (an exception is designs where RAM is kept “alive” with a small battery). The second type of memory is **read-only memory** (ROM), which is similar to RAM except that new data cannot be written in; all data in ROM are loaded at the factory and cannot be changed by the computer. This memory does not lose its data when power is turned off and is called **non-volatile memory**. Most microprocessor systems have both RAM and ROM. RAM is used for temporary program storage and as a temporary scratch-pad memory for the CPU. ROM is used to store programs and data that need to be always available. Actually, many computers use an EPROM (erasable programmable read-only memory) or an EEPROM (electrically erasable programmable ROM) instead of a ROM for long-term memory. EPROMs can be erased with a strong UV light and reprogrammed. EEPROMs can be erased and reprogrammed electrically. Disk drives also store digital data but in a form that must be processed before they are accessible to the microprocessor.

The **input/output (I/O)** section of the computer allows it to interface with the outside world. The input section is the conduit through which new programs and data are entered into the computer, and the output section allows the computer to communicate its results. An I/O interface is called a **port**. An input port is a circuit that connects input devices to the computer; examples of input devices are keyboards, sensors, and switches. An output port is a circuit that connects the computer to output devices. Examples of output devices are indicator lamps, actuators, and monitors. Input/output is discussed in more detail in the next section.

Referring again to Figure 2.2, we see that the blocks are connected by three lines labeled address bus, data bus, and control bus. The **address bus** is a group of wires that carries an address (in binary form) from the CPU to the memory and I/O circuits. The

**Figure 2.3**

A section of memory.

Address	Data
1000	00000010
1001	01100100
1002	00000000
1003	01011100
1004	10011010
↓	↓

need for memory to receive addresses has already been discussed, but you may wonder why I/O ports need addresses. It turns out that all I/O ports are assigned addresses and are treated essentially like memory locations by the CPU. The CPU outputs data to the outside world by sending them to a port address. When the circuitry of the designated output port detects its assigned address on the address bus, it opens and allows data to pass from the data bus to whatever is connected to the port. There are two ways that I/O addressing is done. Some microprocessors use what is called **memory-mapped input/output**, where an I/O address is treated just like another memory address. Other microprocessors treat I/O addresses completely separate from memory addresses.

The **data bus** is a group of eight wires that carries the actual numerical data from place to place within the computer. Figure 2.2 shows how the data bus interconnects all blocks. Data flow in both directions on the data bus. For example, input data enter through the input port and proceed through the data bus to the CPU. If the CPU needs to store these data, it will send them back through the data bus to memory. Data to be outputted are sent (by the CPU) through the data bus to the output port. If the data bus connects to all blocks, how do the data know which block to go to? The answer is the address system. For example, when the CPU sends data to memory, it does it in two steps: First, it puts the destination memory address on the address bus; second, it puts the data on the data bus. When the designated memory detects its own address, it “wakes up” and takes the data from the data bus. The other blocks connected to the buses will ignore the whole sequence because they were not addressed. A good analogy here is the phone system, where the phone number is analogous to the memory address. Even though thousands of phones may be connected to the system, when you dial a number, only the designated phone rings. The beauty of the bus system is that it is expandable. Memory or addressable I/O units can be added to the system by simply connecting them to the buses.

The **control bus** (see Figure 2.2) consists of timing and event-control signals from the CPU. These signals are used to control the data flow on the data bus. For example, one of the control signals is the **read/write (R/W) line**. This signal informs the memory if the CPU wishes to read existing data out of memory or write new data into memory. Non-memory-mapped machines have a *memory-I/O control line*. This signal informs the system if the current data exchange involves memory or an I/O port. In general, the control bus is not as standardized as are the address and data buses.

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## 2.2 INTRODUCTION TO MICROPROCESSOR OPERATION

The microprocessor works by executing a program of instructions. Creating the program is similar in concept to programming in BASIC, C, or any other high-level computer language. Each type of microprocessor has its own **instruction set**, which is the set of commands that it was designed to recognize and obey. Microprocessor instruc-



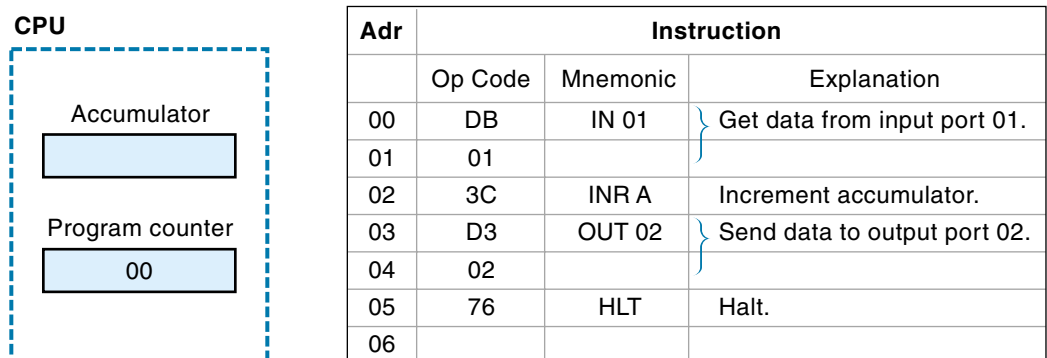
tions are very elemental and specific, and it usually takes more than one to accomplish what a single, high-level language instruction would. Many microprocessor instructions simply move data from one place to another within the computer; others perform mathematical or logic operations. Still another group of instructions control program flow, such as jumping forward or backward in the program. Each instruction in the instruction set is assigned its own unique **operation code**, (which is typically 8 bits long and referred to as the **op-code**). The CPU uses this 8-bit number to identify the instruction.

All microprocessors have at least one **accumulator** [Figure 2.4(a)], which is a data-holding register in the CPU. The accumulator acts as a “staging area” for data. It is common for data coming to the CPU to go first to the accumulator, where it can be operated on. Similarly, most data leaving the CPU exits from the accumulator. Mathematical operations usually store the result in the accumulator. Many of the instructions involve the accumulator in one way or another.

A **machine language** program is a list of instructions (in op-code form) for the microprocessor to follow. Before the program can be executed, it must first be loaded sequentially into memory. The op-code for the first instruction is loaded at the first address location, the op-code for the second instruction is loaded next in line, and so on.

Figure 2.4(b) shows a section of memory with a short program loaded in. The program listing includes the address, op-code, mnemonic, and a brief explanation. (A **mnemonic** is an English abbreviation of an instruction. A program listing using only mnemonics is called **assembly language**.) The program in Figure 2.4(b) directs the CPU to get 1 byte of data from input port 01, add 1 to it, and send the result to output port 02. Before execution can start, the address of the first instruction must be loaded into the program counter. The **program counter** is a special address-storage register that the CPU uses to keep track of where it is in the program, much like a bookmark. The program counter always holds the address of the next instruction to be executed. Once the

**Figure 2.4**  
The CPU uses an accumulator and a program counter to execute a simple program.



Op codes are for Intel 8085 in hexadecimal. In this case, hexadecimal is used as a shorthand form of binary.

(a) Microprocessor registers

(b) Example microprocessor program

microprocessor is activated, execution of the program is completely automatic. The execution process is a series of **fetch-execute cycles**, whereby the microprocessor first fetches the instruction from memory and then executes it. The following are the specific steps the microprocessor would go through to execute the program of Figure 2.4(b):

1. The microprocessor fetches the first instruction from memory. It knows where to find the instruction because its address is in the program counter.
2. Once in the CPU, the op-code is decoded to see which instruction it is, then the proper hardware is activated to execute this instruction. In the example program of Figure 2.4(b), the first instruction (IN 01) is 2 bytes long. The first byte of the instruction is the op-code, telling the CPU to input data from a port. The second byte of the instruction tells the CPU *which* port to read from. Execution of this instruction causes data from input port 01 to travel along the data bus to the accumulator. Also, the program counter advances to 02 (the address of the next instruction). Execution of the first instruction is now complete.
3. The next fetch-execute cycle starts, this time fetching the instruction from address 02. The new instruction (INR A) is “increment the accumulator,” so the accumulator is sent to the ALU to be incremented (add 1) and the result put back in the accumulator. The program counter advances to 03, which is the address of the next instruction.
4. The next fetch-execute cycle starts, this time fetching the instruction from address 03. The instruction (OUT 02) is executed, causing the accumulator data to be sent to output port 02.
5. The final instruction is fetched. It is a “halt,” which causes the microprocessor to cease operating and go into a wait mode.

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## 2.3 INTERFACING TO A MICROPROCESSOR CONTROLLER

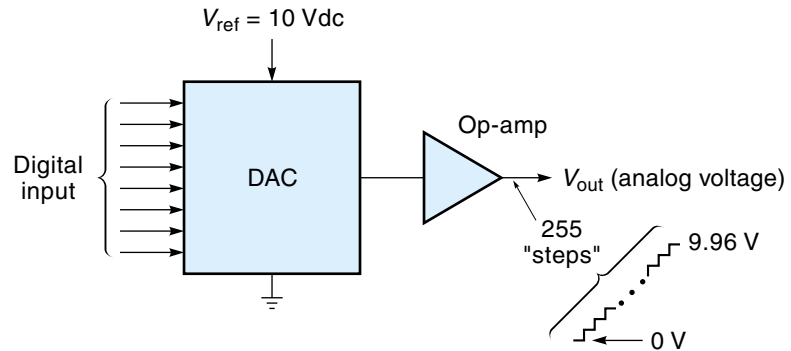
An important part of any control system is the link between the controller and the real world. For a digital controller, data enter and exit through a parallel interface or through a serial interface. Both data formats are discussed next.

### The Parallel Interface

The **parallel interface** transfers data 8 bits (or more) at the same time, using eight separate wires. It is essentially an extension of the data bus into the outside world. The parallel interface is ideal for inputting or outputting data from devices that are either on or off. For example, a single limit switch uses only one input bit, and an on-off signal to a motor requires only one output bit. These 1-bit signals are called **logic variables**, and eight such signals can be provided from a single (8-bit) port. This concept will be expanded on later in this section.

**Figure 2.5**

A digital-to-analog converter (DAC) block diagram.



In other applications, the controller may use a parallel interface to connect to an analog device—for example, driving a variable-speed DC motor. In such a case, the binary output of the controller must first be converted into an analog voltage before it can drive the motor. This operation is performed by a special circuit called a digital-to-analog converter.

### Digital-to-Analog Conversion

The **digital-to-analog converter** (DAC) is a circuit that converts a digital word into an analog voltage. It is not within the scope of this text to describe the internal workings of the DAC, but a general understanding of the operating parameters is appropriate.

Figure 2.5 shows the block diagram of a typical 8-bit DAC. The input is an 8-bit digital word. The output is a current that is proportional to the binary input value and must be converted to a voltage with an op-amp. A stable reference voltage ( $V_{ref}$ ) must be supplied to the DAC. This voltage defines the maximum analog voltage—that is, for a digital input of 11111111,  $V_{out}$  is essentially  $V_{ref}$ . If the input is 00000000, the  $V_{out}$  will be 0 Vdc. For all values in between, the output voltage is a linear percentage of  $V_{ref}$ . Specifically, the output voltage for any digital input (for the 8-bit DAC) is

$$V_{out} = \frac{\text{input} \times V_{ref}}{256} \quad (2.1)$$

where

$V_{out}$  = DAC output analog voltage

input = decimal value of the binary input

$V_{ref}$  = reference voltage to the DAC

### EXAMPLE 2.2

An 8-bit DAC has a  $V_{ref}$  of 10 V. The binary input is 10011011. Find the analog output voltage.

**SOLUTION**

The binary input of 10011011 has a decimal value of 155. Applying Equation 2.1, we can calculate the analog output voltage:

$$V_{\text{out}} = \frac{\text{input} \times V_{\text{ref}}}{256} = \frac{155 \times 10 \text{ V}}{256} = 6.05 \text{ V}$$

Therefore, 6.05 V is the voltage we would expect on the analog output pin. [It is interesting to note that if the input were all 1s (which is a decimal value of 255), the output would be  $(255/256) \times 10 \text{ V} = 9.96 \text{ V}$ , not 10 V as you might expect. This is a characteristic of the DAC.]

An important consideration of digital-to-analog conversion is **resolution**. The resolution of a DAC is the worst case error that is introduced when converting between digital and analog. This error occurs because digital words can only represent discrete values, as indicated by the stair-step diagram in Figure 2.5. For example, the maximum value of an 8-bit number is 255 decimal, which means there are 255 possible “steps” of the output voltage. The difference between steps is the value of the least significant bit (LSB). Because the smallest increment is one step, the resolution (for 8-bit data) is 1 part in 255, or 0.39%. This resolution is adequate for many applications, but if more is needed, two (or more) 8-bit ports can be used together. Two ports provide 16 bits of data. The maximum decimal value of 16 bits is 65,535. Being able to divide an analog number into 65,535 parts means that each part will be much smaller, so we can more precisely represent that number.

**EXAMPLE 2.3**

A computer uses a DAC to create a voltage that represents the position of an antenna. The antenna can rotate  $180^\circ$  and must be positioned to within  $1^\circ$ . Can an 8-bit port be used?

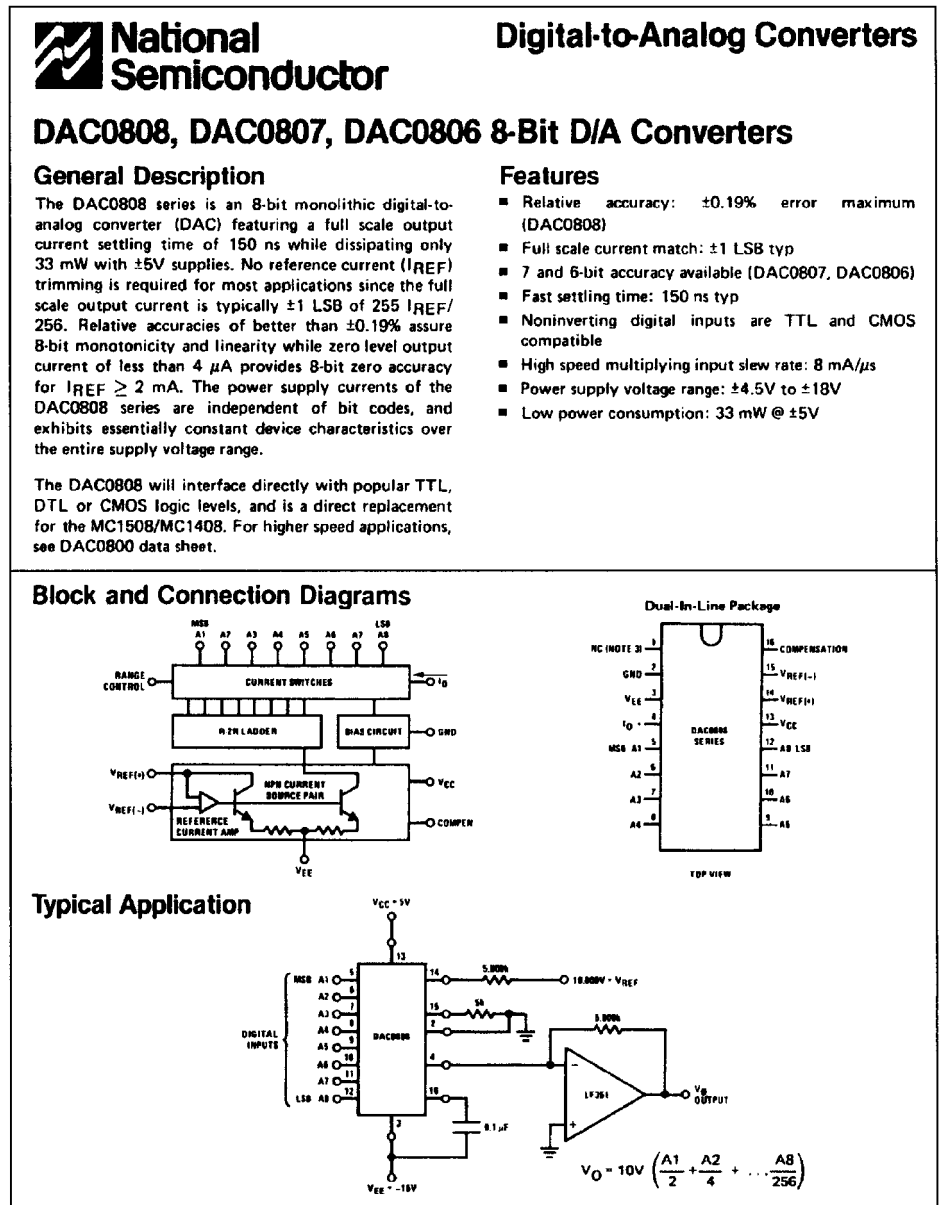
**SOLUTION**

The resolution required is 1 part in 180. Because 8 bits provide a resolution of 1 part in 255, an 8-bit port is certainly adequate. In fact, we have a choice: We could have the  $\text{LSB} = 1^\circ$ , in which case the input values would range from 0 to 180, or we could equate  $180^\circ$  with 255, which makes the  $\text{LSB} = 0.706^\circ$ . The latter makes maximum use of the 8 bits to give a better resolution, but if the system really doesn't need it, the clear, simple relationship of  $\text{LSB} = 1^\circ$  is desirable.

Figure 2.6 shows a data sheet for an 8-bit DAC (DAC0808). This device comes as a 16-pin DIP (dual in-line package) and uses an external op-amp (such as the LF 351), two resistors, and a capacitor to complete the circuit. It requires plus and minus power-supply voltages. The time to complete a conversion is a fast 150 ns (nanoseconds). (The circuit shown in Figure 2.6 has a  $V_{ref}$  of 10 Vdc.)

**Figure 2.6**

The data sheet for the DAC0808, an 8-bit digital-to-analog converter. (Courtesy of National Semiconductor Corp.)



### Analog-to-Digital Conversion

An **analog-to-digital converter** (ADC) is a circuit that converts an analog voltage into a digital word. A typical ADC consists of a single IC with a few support components. Analog-to-digital conversion is a more complicated process (than for the DAC), and the hardware requires some conversion time, which is typically in the microsecond range. The conversion time required depends on the type of ADC, the applied clock frequency, and the number of bits being converted. Figure 2.7 shows a block diagram for an 8-bit ADC. The input  $V_{in}$  can be any voltage between 0 V and  $V_{ref}$ . When  $V_{in}$  is 0 Vdc, the output is 00000000; when  $V_{in}$  is  $V_{ref}$ , the output is 11111111 (255 decimal). For input voltages between 0 and  $V_{ref}$ , the output increases linearly with  $V_{in}$ ; therefore, we can develop a simple ratio for the ADC:

$$\frac{\text{output}}{V_{in}} = \frac{255}{V_{ref}} \quad (\text{for 8 bits})$$

Solving for output gives the following:

$$\text{Output} = \frac{V_{in} \times 255}{V_{ref}} \quad (2.2)$$

where

output = decimal output value of an 8-bit ADC

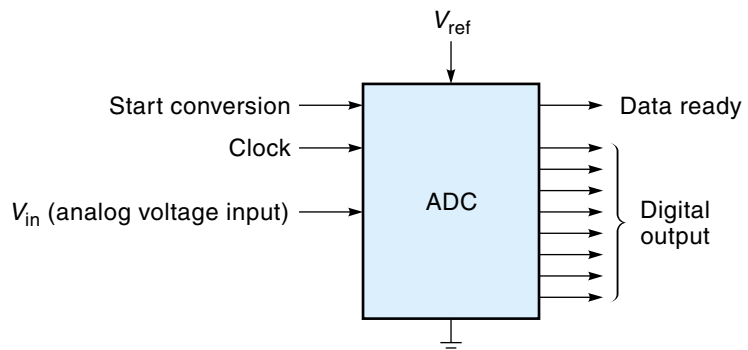
$V_{in}$  = analog input voltage to the ADC

$V_{ref}$  = ADC reference voltage

To start the conversion process, a start-conversion pulse is sent to the ADC. The ADC then samples the analog input and converts it to binary. When completed, the ADC activates the data-ready output. This signal can be used to alert the computer to read in the binary data.

**Figure 2.7**

An analog-to-digital converter (ADC) block diagram.



**Figure 2.8**

The data sheet for the ADC0804, an 8-bit analog-to-digital converter. (Courtesy of National Semiconductor Corp.)



**National Semiconductor**

**Analog-to-Digital Converters**

**ADC0801, ADC0802, ADC0803, ADC0804 8-Bit  $\mu$ P Compatible A/D Converters**

**General Description**

The ADC0801, ADC0802, ADC0803, ADC0804 are CMOS 8-bit, successive approximation A/D converters which use a modified potentiometric ladder—similar to the 256R products. They are designed to meet the NSC MICROBUS™ standard to allow operation with the 8080A control bus, and TRI-STATE® output latches directly drive the data bus. These A/Ds appear like memory locations or I/O ports to the microprocessor and no interfacing logic is needed.

A new differential analog voltage input allows increasing the common-mode rejection and offsetting the analog zero input voltage value. In addition, the voltage reference input can be adjusted to allow encoding any smaller analog voltage span to the full 8 bits of resolution.

**Features**

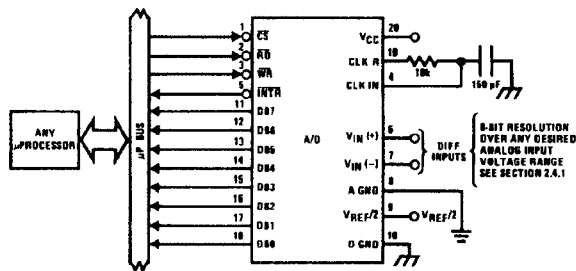
- MICROBUS (8080A) compatible—no interfacing logic needed
- Easy interface to all microprocessors, or operates “stand alone”

- Differential analog voltage inputs
- Logic inputs and outputs meet T<sup>2</sup>L voltage level specifications
- Works with 2.5V (LM336) voltage reference
- On-chip clock generator
- 0V to 5V analog input voltage range with single 5V supply
- No zero adjust required
- 0.3” standard width 20-pin DIP package

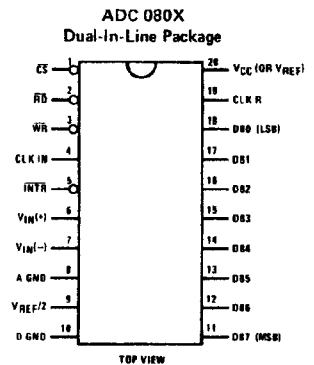
**Key Specifications**

- Resolution 8 bits
- Total error  $\pm 1/4$  LSB,  $\pm 1/2$  LSB and  $\pm 1$  LSB
- Conversion time 100  $\mu$ s
- Access time 135 ns
- Single supply 5 V<sub>DC</sub>
- Operates ratiometrically or with 5 V<sub>DC</sub>, 2.5 V<sub>DC</sub>, or analog span adjusted voltage reference

**Typical Applications**



**Connection Diagrams**



**EXAMPLE 2.4**

An 8-bit ADC has a  $V_{\text{ref}}$  of 7 Vdc; the analog input is 2.5 Vdc. What is the binary output of the ADC?

**SOLUTION**

The output is an 8-bit word that has a maximum decimal value of 255 (decimal) when  $V_{\text{in}} = V_{\text{ref}}$ . Therefore, an analog input voltage ( $V_{\text{in}}$ ) of 7 Vdc would be converted to 255 decimal. Using this set of I/O data, we can develop a ratio and then use that to find the output for the specific input of 2.5 Vdc:

$$\frac{\text{output}}{V_{\text{in}}} = \frac{255}{7 \text{ Vdc}}$$

Solving for output gives the following:

$$\text{Output} = \frac{2.5 \text{ Vdc} \times 255}{7 \text{ Vdc}} = 91$$

The result is 91 decimal = 01011011 binary. This is the output that would appear on the eight output lines of the ADC.

Figure 2.8 shows a data sheet for an 8-bit ADC (ADC0804). Packaged as a 20-pin DIP, this device can operate on a single 5-Vdc power supply and requires an external resistor and capacitor to complete the ADC circuit. The start-conversion pulse is applied to pin 3 (WR), and the data-ready signal comes from pin 5 (INTR). This particular ADC can be connected in a free-running mode where it performs one conversion after the other as fast as it can. Notice also that the pin labeled  $V_{\text{ref}/2}$  (pin 9) must be set at half of the actual  $V_{\text{ref}}$ . For example, if the requirements call for an analog voltage range of 0-5 Vdc, then pin 9 would be set to 2.5 Vdc. The time to complete a conversion is approximately 100  $\mu\text{s}$  (micro-seconds), making it almost 700 times slower than the DAC0808 discussed earlier.

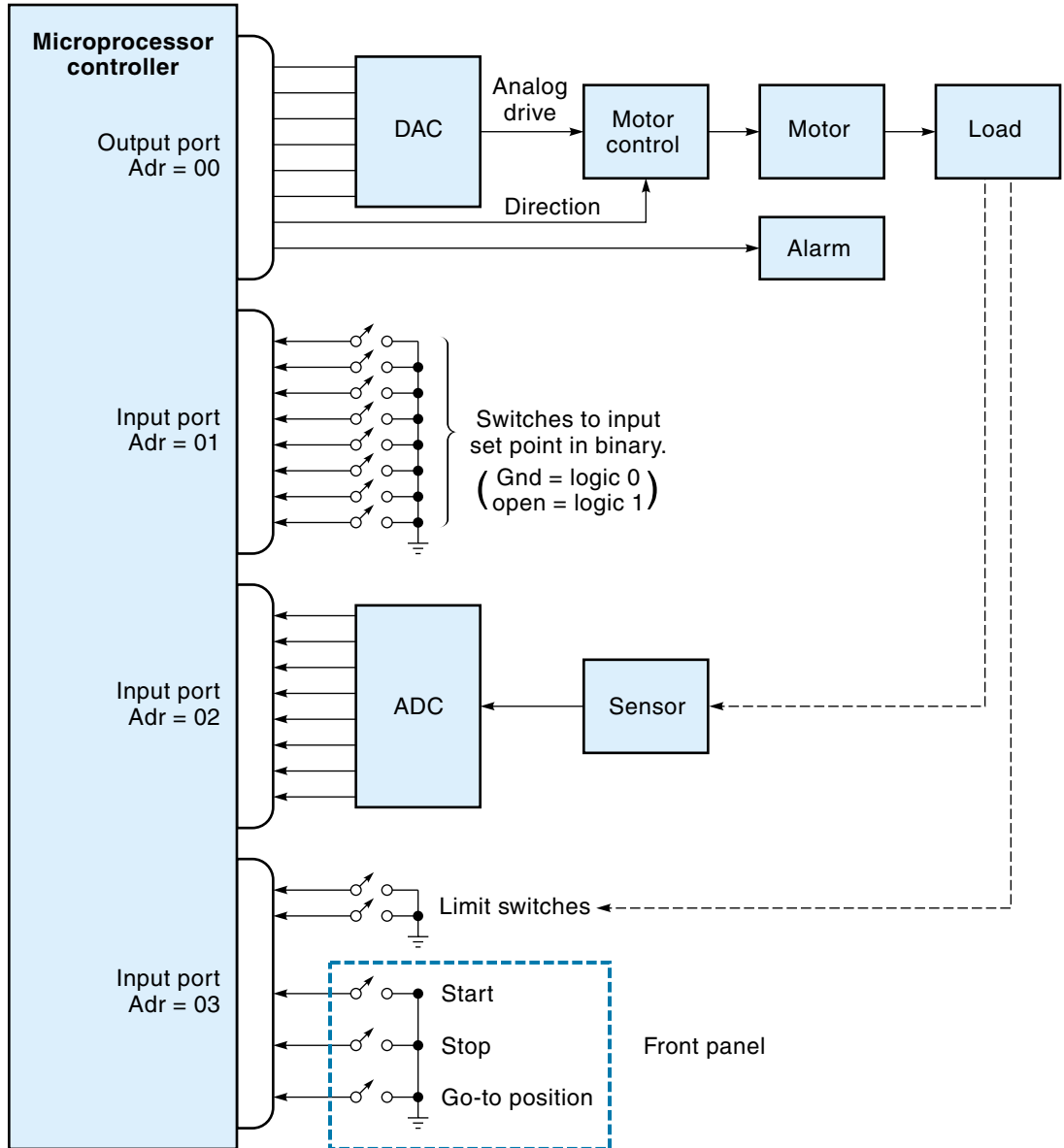
***A Control System Using Parallel Ports***

Figure 2.9 shows a position control system using a microprocessor-based controller with parallel ports. This particular system has one output port and three input ports (each port has its own address). The output port is partitioned: Six bits are converted in a DAC to provide the analog motor-drive signal, the seventh bit specifies motor direction (1 = clockwise, 0 = counterclockwise), and the eighth bit turns on an audio alarm if some emergency situation is detected. The first input port inputs the set-point data, the second inputs the ADC data from the sensor, and the third inputs various 1-bit logical variables. In this case, the system has three front-panel switches as well as two limit



**Figure 2.9**

A control system using parallel interface.



switches. The limit switches are used as a “back up” to detect it if the load has gone out of its designated range.

Operation of the system proceeds as follows: The controller inputs the data from port 03 to determine if the start (or stop) button has been pressed. If the start button has been pressed, then the set point is read in from port 01 and the digitized sensor data is read in from port 02. Based on its control strategy, the controller outputs to port 00 a binary word representing the motor-control voltage. This digital data is converted to an analog voltage with the DAC. This entire sequence is repeated over and over until the stop button is pushed.

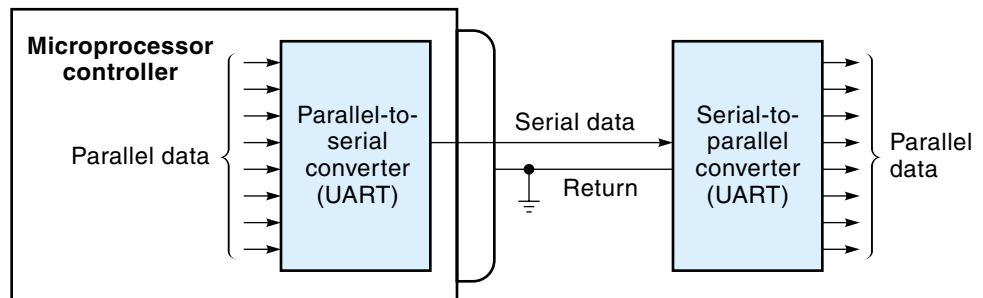
### The Serial Interface

In a **serial interface**, the data are sent 1 bit after the other on a single wire. There are a number of good reasons for doing this. First, the cabling is simpler because only two wires are needed (at a minimum), those being “data” and “return.” Second, shielding a small group of wires, which is often necessary in an electrically noisy industrial environment, is easier. Third, serial data can make use of existing single-channel data lines such as the telephone system (which may require using a modem). For these reasons, serial data transfer is usually recommended for distances greater than 10-30 ft.

Because data always exist in a parallel form inside the computer, it must be converted to serial data before coming out the serial port. This is accomplished with a special parallel-to-serial converter IC called a **universal asynchronous receiver transmitter (UART)**. On the other end of the line, a receiver must convert the serial data back into parallel data, which is done with another UART. Figure 2.10 shows the basic serial data circuit.

Serial data are classified as being either synchronous or asynchronous. *Synchronous data* require that the data bytes be sent as a group in a “package.” It is used in sophisticated communication systems that move a lot of data and will not be further discussed here. *Asynchronous data* transfer is the more common (but slower) type of serial transfer and allows for individual bytes to be sent when needed.

**Figure 2.10**  
Components in a  
serial interface  
circuit.



**Figure 2.11**

Serial data format for the binary word 10110010 (with even parity).

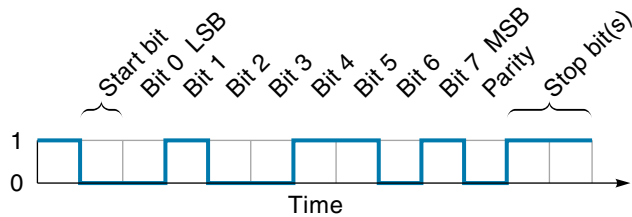


Figure 2.11 shows the standard format for asynchronous serial data. First, a start bit is sent, then the data (LSB first), then a parity-error checking bit, and finally the stop bit(s). Some variation is allowed to this format, but both transmitter and receiver must use the same format. The other important parameter in serial transmission is the number of bits sent per second (frequently called the **baud** rate, although the term is technically incorrect in most cases). Standard bit rates are 300 bps (bits per second), 1200 bps, 2400 bps, 9600 bps, 14,400 bps, 28,800 bps, 33,600 bps, and 57,600 bps. Serial data transmission is much slower than parallel transmission. At 300 bps, it takes almost 37 ms to transmit 1 byte of data, compared to less than a microsecond for parallel—this is thousands of times slower. Still, for many applications, particularly process control, the longer data-transfer times are not a problem.

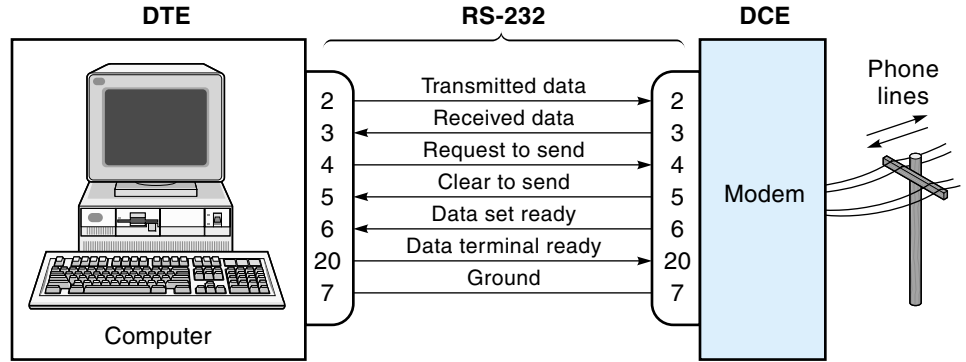
### RS 232

In order to make the serial interface practical, a set of specifications called the RS-232 standard was established. Officially, the **RS-232 standard** specifies the serial data interface between **data terminal equipment** (DTE) and **data communication equipment** (DCE). A common application of RS-232 is the interface between a PC and the modem, in which case the computer is the DTE and the modem is the DCE [see Figure 2.12(a)] A **modem** is a device that converts digital data into audio tones so that it can be transmitted over the telephone lines. As shown in Figure 2.12, the RS-232 interface consists of seven signals; the serial data is sent on pin 2 and received on pin 3; the other signals, such as “Request to send” and Clear to send,” are used to confirm that the two units are ready to communicate. The RS-232 standard specifies connector types, signal names, pin numbers, and voltages. In practice, the RS-232 standard can be applied to any serial interface as long as one unit acts as a DTE and the other as a DCE. If two DTE units need to interface with each other—for example, a PC to a PC—a special cable called a **null modem** or **crossover cable** is used. RS-232 is commonly used in the control field when two units need to exchange data—for example, to connect a PC to a local control unit for the purpose of downloading a new control program, as illustrated in Figure 2.12(b).

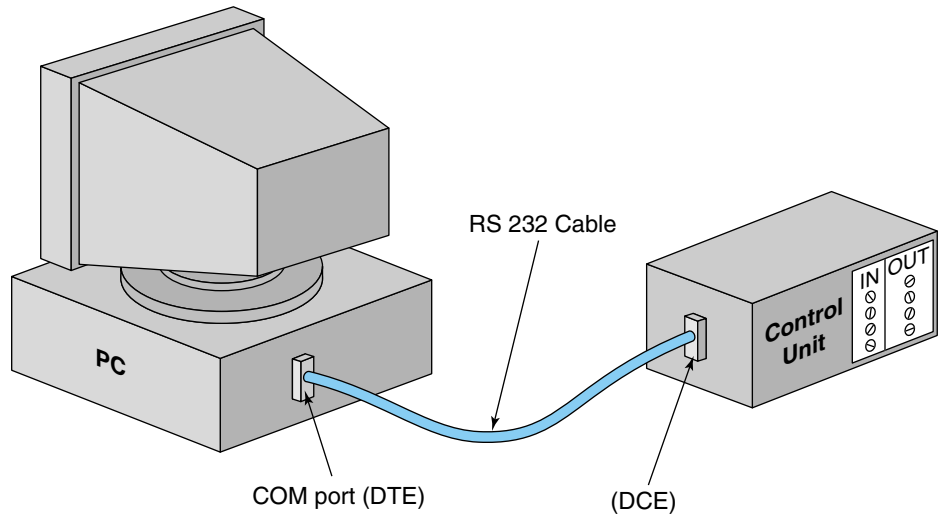
RS-232 serial data transfer is somewhat more complicated than parallel data transfer, but it offers advantages such as two-wire communications and a universally accepted interface. The hardware to handle serial data is standardized, readily available, and reliable.

**Figure 2.12**

The RS-232 serial interface.



(a) Interface between data terminal equipment (DTE) and data communication equipment (DCE). Serial data are transferred on pins 2 and 3; the other signals control the flow of data.



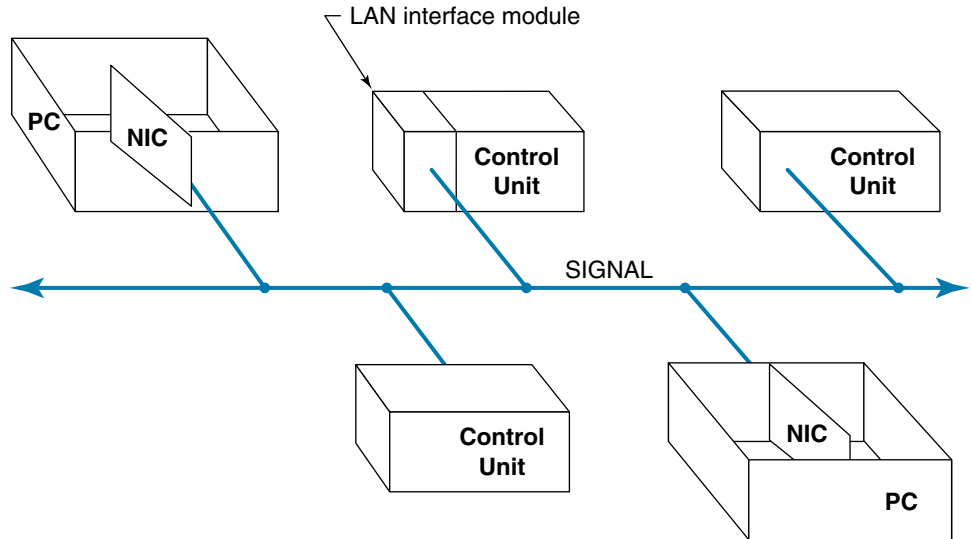
(b) Using an RS-232 serial cable to connect a PC to a controller

### Networking

Probably the most common use of serial data is in networking. More and more, networks are being used to interconnect all the units and devices in the control system. Network cabling differs depending on the type of **local area network (LAN)**, but most use the generalized bus system diagrammed in Figure 2.13. Typically, each unit on the net has a unique address number and also address detection circuitry. When one unit wants to talk to another unit, it first broadcasts the address of the unit it wants to talk to (serially, of course, on the signal wire) and then sends the data (seri-

**Figure 2.13**

Concept of local area network (LAN) interconnections for bus topology.



ally), which consists of some number of bytes. All units on the net will receive the address, but only the intended receiver will activate and then read in the data. The interface between the network cable and the PC is done through a commercially available interface expansion card called a network interface card (NIC). Other devices on the net, such as control units, would require a special interface circuit, which may be built in or available as an external module. Control system networks are discussed further in Chapter 12.

## 2.4 INTRODUCTION TO CONTROLLER PROGRAMMING

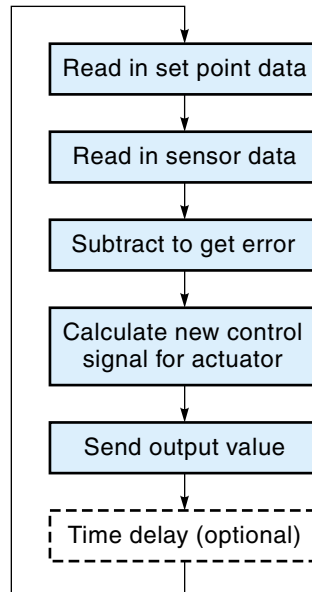
It is beyond the scope of this text to present a detailed discussion of how to program a microprocessor in machine language. Still, it is useful to investigate in a general way what the software must do. A digital controller is a computer operating in **real time**. This means that *the program is running all the time—repeatedly taking in the newest sensor data and then calculating a new output for the actuator*.

The basic structure of a controller program is a *loop*. In a loop structure, the same sequence of instructions is executed over and over again, and each pass through the loop is called an **iteration**, or **scan**. Figure 2.14 shows a generalized controller program, and an explanation of the program follows:

1. The program reads in the set-point data (recall that the set point is the desired position of the controlled variable). This data could be read in from an input port or from memory.

**Figure 2.14**

A generalized controller program.



2. The program directs the computer to read (from a sensor) the actual value of the controlled variable.
3. The actual data are subtracted from the set point to get the error.
4. Based on the error data, the computer calculates a new actuator control signal.
5. The new output is sent to the actuator.
6. The program loops back to step 1 and starts over again.

The time it takes for the computer to execute one pass through the loop determines the time interval between input readings (known as the **sampling rate**). If this interval is too long, the computer may not get an accurate picture of what the controlled variable is really doing (see Chapter 11 for a discussion of aliasing). Execution of the loop can be accelerated by specifying a faster computer or streamlining the program. In other situations, the computer must pause and wait. For example, a pause might be inserted to give an operator time to make some adjustment or to allow time for a motor to “spin down.” This is done by inserting time-delay loops in the program. A **time-delay loop** is simply a do-nothing, “wheel-spinning” loop where the computer is instructed to count up to some large number. Using this technique, we can make the program pause for any length of time—from a few microseconds to hours. If a time-delay loop is inserted in the main program loop (as shown in Figure 2.14), the effect is to slow the cycle time for the main loop. This is sometimes done to force matching of the sample rate to some predetermined value.

At one time, people thought that the best and most efficient microprocessor programs were those written directly in assembly language—that is, the programmer would directly select the machine language instructions. Today, sophisticated programs (called

*compilers*) can convert a program written in a high-level language, primarily C, into very efficient machine language. High-level languages use English-sounding words and a set of powerful commands to specify simple and complicated programming operations with a minimum of instructions. Using a high-level language to write programs for a microprocessor offers big advantages, such as more compact program listing, ease of writing equations, and more comprehensible documentation. Also, programs written in a high-level language can be compiled to run on any model of microprocessor.

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## 2.5 MICROPROCESSOR-BASED CONTROLLERS

### Single-Chip Microcomputers (Microcontrollers)

A microprocessor by itself is not a computer. To be functional, the microprocessor must be connected to other integrated circuits that provide the memory and I/O capability. A microcontroller is a computer on a single IC, designed specifically for control applications. It consists of a microprocessor, memory (both RAM and ROM), I/O ports, and possibly other features such as timers and ADCs/DACs. Having the complete controller on a single chip allows the hardware design to be simple and very inexpensive. Microcontrollers are showing up increasingly in products as varied as industrial applications, home appliances, and toys. In such uses as these, they are called **embedded controllers** because the controller is located physically in the equipment being controlled.

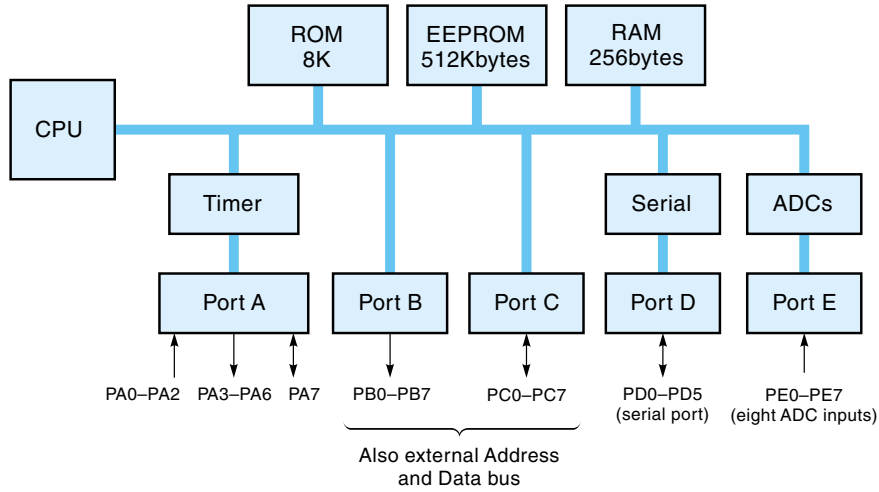
The main difference between microprocessors and microcontrollers is that microprocessors are being designed for use in microcomputers where greater speed and larger word size are the driving requirements, whereas microcontrollers are evolving toward reduced chip count by integrating more hardware functions on the chip. Most control applications do not need the 32-bit word size and 500-MHz (megahertz) speed of the newer microprocessors. Eight or 16 bits and 1 MHz will work just fine in many applications, *and* the single-chip microcontroller costs much less.

Another difference between microprocessors and microcontrollers concerns the instruction set. The microprocessor tends to be rich in instructions dealing with moving data into and out of memory. The microcontroller has fewer memory-move instructions and more bit-handling instructions. The reason for the lack of memory-move instructions is that the microcontroller typically has only a small amount of RAM, which it uses only as a “scratch pad.” The additional bit-handling instructions were included because they are so useful in control system applications. For example, in a control system, each separate bit of a parallel output word might control a different device, such as a motor or indicator light. The bit-handling instructions allow the software to turn one device easily on or off without affecting the others.

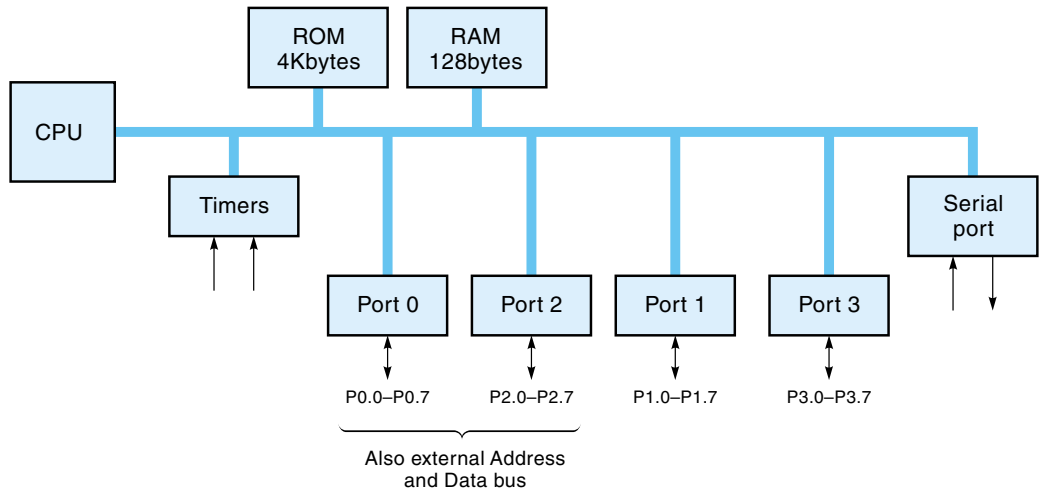
The Motorola 68HC11 is a popular 8-bit microcontroller that has 256 bytes of RAM, 8K of ROM, and 512K bytes of EPROM (see Figure 2.15a). It also has five 8-bit ports with built-in serial data transfer and ADC capability. Another common 8-bit

**Figure 2.15**

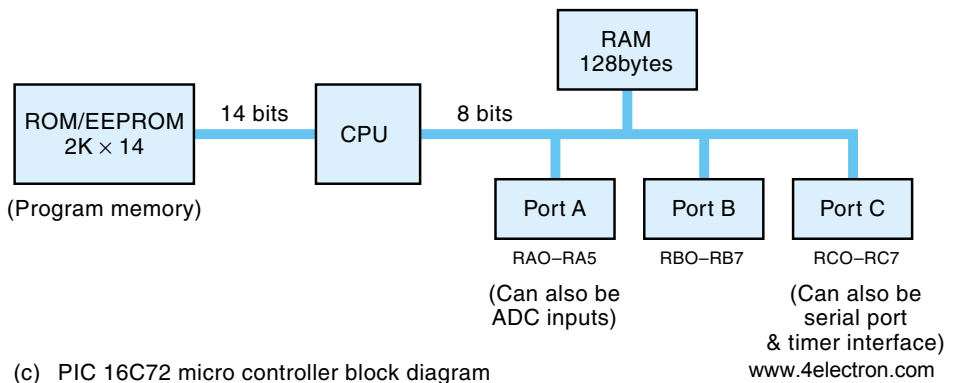
Block diagrams of microcontrollers.



(a) Motorola 68HC11 microcontroller block diagram



(b) Intel 8051 microcontroller block diagram



(c) PIC 16C72 micro controller block diagram



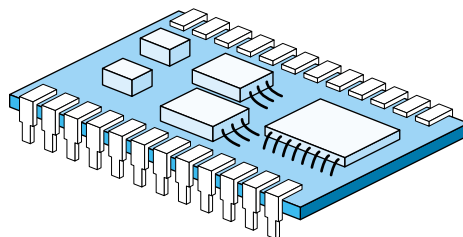
microcontroller is the Intel 8051, which has 128 bytes of RAM and 4K bytes of ROM, four parallel data ports, and a serial port (see Figure 2.15b). For control applications, these hardware arrangements usually are adequate: ROM is used to store the control program, and RAM is used as data registers and a “scratch pad.” The I/O signal lines can usually be connected directly to the microcontroller without additional port circuitry. Software is typically written in C++ or some other language (including assembly language) and then converted into machine language with a compiler or assembler program. The machine language program would then be loaded into the microcontroller’s ROM or EPROM.

Another popular microcontroller is the PIC from Microchip Technology. For example, the PIC16Cxx family of 8-bit microcontrollers is a low-cost, versatile product that has found wide acceptance [see Figure 2.15(c)]. There are a wide range of options, including ROM, EPROM, EEPROM, ADCs, Timers, and serial ports. The PIC uses a slightly different architecture from the 68HC11 and 8051 in that the ROM (or EPROM) that contains the program connects to the CPU with its own 14-bit bus, whereas the regular data bus is 8 bits. Allowing 14 bits for the program memory means that all instructions are just one word long. The device has three I/O ports, but many of the I/O bits can be used in different ways (such as for an on/off switch or ADC input), depending on how they are programmed.

Finally, another product called the BASIC Stamp from Parallax Inc. is usually considered a microcontroller, although it is actually a very small circuit board with a few ICs and pins. The whole circuit board plugs into an IC socket, as though it were an IC (see Figure 2.16 ). What makes the BASIC Stamp somewhat unique is that it has an on-board BASIC program interpreter. A program can be written in BASIC on a PC and then directly downloaded into the Stamp’s EEPROM through a RS-232 port. No assembler or compiling operation is required. There are now other microcontrollers on the market that can be programmed in BASIC.

In summary, a wide variety of microcontrollers are available. At the low end are the 4-bit models, which are more than adequate for appliances and toys. These tend to be large-volume, low-cost applications. Eight-bit microcontrollers (such as the 68HC11 and 8051 mentioned earlier) are very popular because 8 bits turn out to be a convenient size for both numeric and character data. At the high end, 16- and 32-bit microcontrollers are available for control systems requiring sophisticated, high-speed

**Figure 2.16**  
BASIC Stamp  
module (BS2-IC).



calculating power for such applications as complicated servomechanisms, avionics, or image processing.

### Single-Board Computers

**Single-board computers** are off-the-shelf microprocessor-based computers built on a single printed-circuit card (Figure 2.17). They come in many configurations, but in general they use a standard microprocessor such as the Zilog Z80, the Intel x86 family, the Motorola 68000, or a microcontroller. They also include memory ICs (both RAM and ROM), I/O capability, and perhaps special interface circuits such as ADCs or DACs. Single-board computers are manufactured by major microprocessor producers such as Intel and Motorola as well as many other smaller companies. Some single board computers are designed to plug into a PC as an expansion card. The obvious advantage of using a ready-made microprocessor board is that it eliminates design- and board-testing time. This is particularly important in small-volume production or one-of-kind systems.

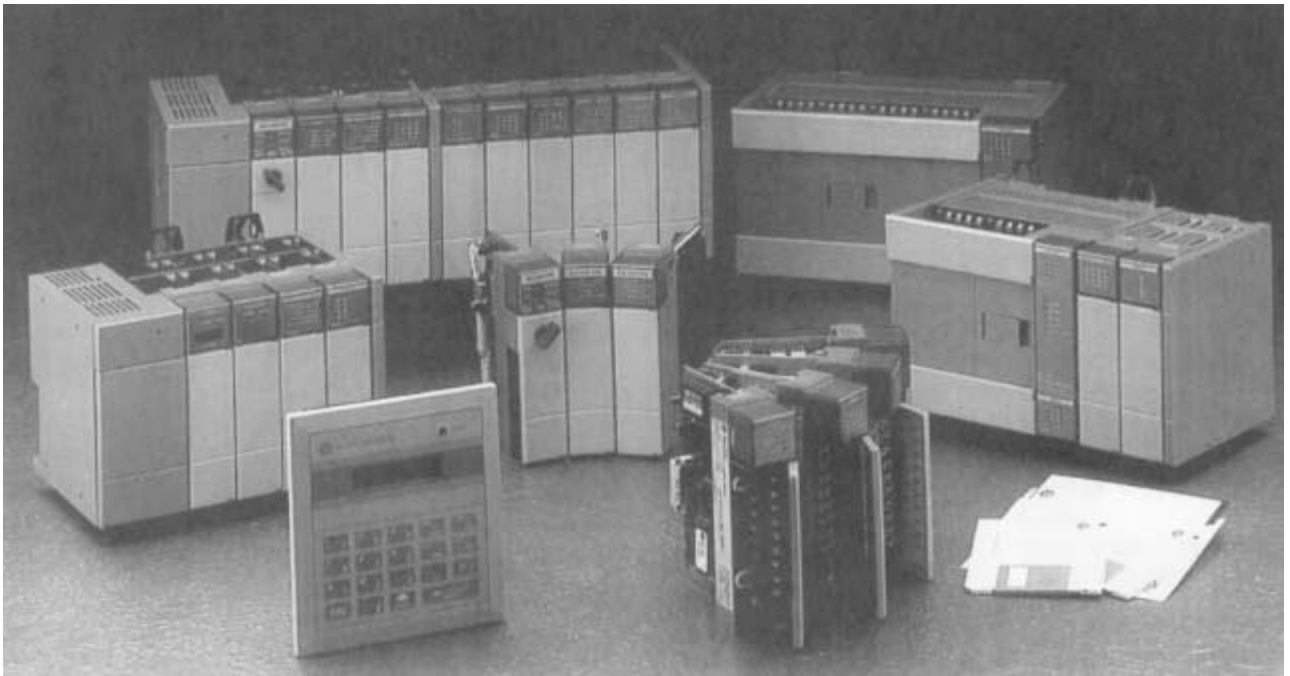
### Programmable Logic Controllers

A **programmable logic controller** (PLC) is a self-contained microprocessor-based unit, designed specifically to be a controller. The PLC includes an I/O section that can

**Figure 2.17**  
A single-board computer.  
(Courtesy of Vesta Technology, Inc.)



interface directly to such system components as switches, relays, small motors, and lights. Developed in the late 1960s to replace relay logic controllers, PLCs have evolved to be able to handle sophisticated motion control applications. PLCs come in various sizes and capabilities; Figure 2.18 shows a selection of PLCs. The big difference between PLCs and the other devices discussed in this section is that the PLC has the microprocessor, ports, and power supply built into a package that has been ruggedized for an industrial environment. Installation is very easy because in many cases the sensors and actuators can be connected directly to the PLC. Once installed, the microprocessor program is **downloaded** into the PLC from some source such as a personal computer. The PLC manufacturer usually supplies software to facilitate the programming operation. This software allows the user to write a program with line-by-line instructions, or it can convert a relay logic-wiring diagram (ladder diagram) directly



**Figure 2.18**

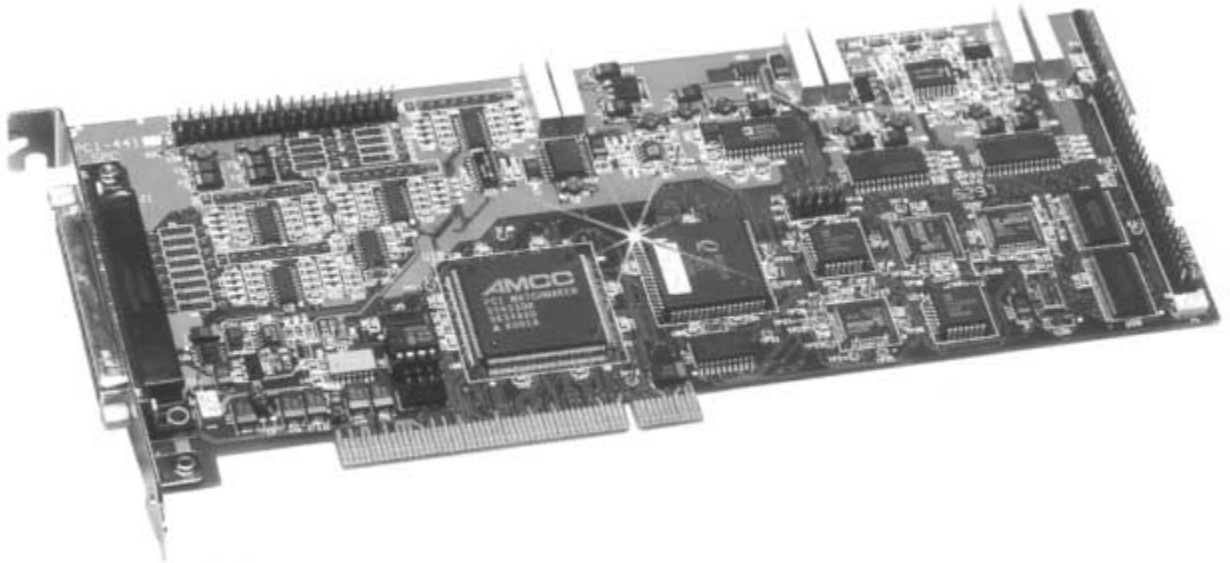
Programmable logic controllers. (Allen-Bradley products courtesy of Rockwell Automation).

into a PLC program. Multiple PLCs in a plant can be networked so the individual units can be monitored and programmed from a single station. This is a form of distributed computer control (DCC) discussed in Chapter 1. PLCs are discussed in detail in Chapter 12.

### Personal Computers Used in Control Systems

The availability of relatively low-cost, off-the-shelf **personal computers** (PCs) has made them an attractive alternative for small, one-of-kind control applications. Control system software packages are commercially available for the PC that run under DOS and Windows. These programs are adaptable and allow the user to tailor the software to fit the control application, essentially turning a PC into a PLC (although not as rugged). Most of these packages use interactive graphics to link animation with changing process values. Some programs have provisions to mathematically simulate the process being controlled to help optimize the controller coefficients.

A standard PC comes with **expansion slots**, which are circuit-card connectors emanating from the *motherboard* (main board) of the computer. **Expansion cards** plug into these slots and form a bridge between the computer and the outside world. Many different types of interface cards are available, such as I/O serial and parallel data ports,



**Figure 2.19**

Multi-function I/O board, includes ADC, DAC, and digital I/O. (Courtesy of Omega Engineering, Inc.)

ADCs, DACs, and computer-controlled output relays, to name a few. Figure 2.19 shows an example of an interface expansion card.

Historically, data-acquisition and control functions were kept separate. Controllers ran the process, and other instruments measured and recorded the result. The concept of having a single PC perform both tasks seems logical; after all, the PC can use its computing ability first as the controller and then tabulate system performance data. These data can be stored on disk and/or displayed on the monitor.

A potential problem may arise because the controller must operate in real time. If a computer is to control a process and monitor it at the same time, the data-reduction process must not take so long as to interfere with the control duties; a control response can't wait. One way to overcome this problem is to divide the control and data-acquisition tasks among multiple processors. Using the PC as the master computer, a separate microprocessor on an expansion card can perform data collection uninterrupted. One type of I/O controller card has slots for three smaller boards. These smaller boards have various combinations of analog and digital I/O ports and counter-timers. Some boards are available with solid-state relays, which can be used to directly control AC and DC motors.

A PC with I/O expansion cards often costs less than a stand-alone computerized control system. The cards do not need a separate enclosure and use the PC's power supply, keyboard for input, and monitor for display. Also, using a standard PC means that programs can be developed on another compatible computer, eliminating process downtime.

Numerous manufacturers are selling rugged PCs that can survive in harsh industrial environments. These computers typically use a membrane-type keyboard (the keyboard appears as one continuous sheet of flexible plastic) and have sealed cases and filters covering the air vents. Some models of these computers are rack-mountable and contain their own battery-backup power supply.

---

## SUMMARY

A microprocessor is a digital integrated circuit that performs the basic operations of a computer. Microprocessors are used extensively as the basis of a digital controller. Digital control systems are advantageous because digital data can be transferred and stored virtually error-free, and the control strategy can be changed by simply reprogramming.

A computer consists of four basic functional units: (1) the CPU (microprocessor), which executes the programmed instructions and performs the calculations; (2) the memory, which stores the program and data; (3) input; and (4) output. Input/output interfaces the computer to the outside world. A microprocessor-based computer interconnects these units with three groups of signals called buses. The address bus carries the address of the data to be processed. The data bus carries the data, and the control bus carries timing and control signals. Computers handle data as groups of binary bits. Many microprocessor-based controllers handle data in 8-bit groups called a byte.

A microprocessor has a set of instructions that it can execute (called the instruction set). Each instruction is identified by a digital code called the operation code (op-code).

A program consists of a list of these op-codes stored in memory. The microprocessor automatically fetches the instructions from memory and executes them, one by one.

A digital controller may have two kinds of data interfaces: parallel and serial. The parallel interface is the most straightforward system, where all 8 bits are sent at the same time on eight separate wires. In the serial interface, data is sent 1 bit after the other on a single wire. Serial data transfer is better for longer distances.

Many control systems use components that require an analog signal interface; therefore, the signals to or from the digital controller must be converted with an ADC (analog-to-digital converter) or a DAC (digital-to-analog converter). Both circuits are available in IC form.

The digital controller program has a standard format. First, it reads the set point and sensor values. Then it subtracts these values to determine the system error. Based on the error value, it next calculates the appropriate actuator response signal and sends it out. Then it loops back to the beginning of the program and executes the same set of instructions over and over.

Microprocessor-based controllers come in a number of standard forms. A microcontroller includes a microprocessor, memory, and input/output all on a single IC. A single-board computer is an off-the-shelf microprocessor-based computer, assembled onto a single printed circuit board. A programmable logic controller (PLC) is a self-contained unit specifically designed to be a controller. A personal computer (PC) is a general-purpose, self-contained computer; however, with the addition of interface expansion cards, a PC becomes a very adaptable and cost-effective controller.

---

## GLOSSARY

**accumulator** A temporary, digital data-storage register in the microprocessor used in many math, logic, and data-moving operations.

**ADC** See **analog-to-digital converter**.

**address** A number that represents the location of 1 byte of data in memory or a specific input/output port.

**address bus** A group of signals coming from the microprocessor to memory and I/O ports, specifying the address.

**ALU** See **arithmetic logic unit**.

**arithmetic logic unit (ALU)** The part of the CPU that performs arithmetic and logical operations.

**analog-to-digital converter (ADC)** A device (usually an IC) that can convert an analog voltage into its digital binary equivalent.

**assembly language** A computer program written in mnemonics, which are English-like abbreviations for machine-code instructions.

**baud** The rate at which the signal states are changing; frequently used to mean “bits per second.”

**bit** The smallest unit of digital data, which has a value of 1 or 0.

**byte** An 8-bit digital word.

**central processing unit (CPU)** The central part of a computer, the CPU performs all calculations and handles the control functions of the computer.

**control bus** A group of timing and control signals coming from the microprocessor to memory and I/O ports.

**crossover cable** *See null modem.*

**CPU** *See central processing unit.*

**DAC** *See digital-to-analog converter.*

**data bus** A group of signals going to and from the microprocessor, memory, and I/O ports. The data bus carries the actual data that are being processed.

**data communication equipment (DCE)** One of two units specified by the RS-232 standard (for serial data transfer); the DCE is usually a modem.

**data terminal equipment (DTE)** One of two units specified by the RS-232 standard (for serial data transfer). The DTE is usually the computer.

**DCE** *See data communication equipment.*

**digital-to-analog converter (DAC)** A circuit that translates digital data into an analog voltage.

**download** To transfer a computer program or data into a computer (from another computer).

**DTE** *See data terminal equipment.*

**embedded controller** A small microprocessor-based controller that is permanently installed within the machine it is controlling.

**expansion card/slot** An expansion card is a printed circuit card that plugs into an expansion slot on the motherboard of a personal computer (PC). The expansion card usually interfaces the PC to the outside world.

**fetch-execute cycle** A computer cycle where the CPU fetches an instruction and then executes it.

**input/output (I/O)** Data from the real world moving in and out of a computer.

**I/O** *See input/output.*

**instruction set** The set of program commands that a particular microprocessor is designed to recognize and execute.

**iteration** One pass through the computer program being executed by the digital controller; each iteration “reads” the set-point and sensor data and calculates the output to the actuator.

**LAN** *See* **local area network**.

**least significant bit (LSB)** The rightmost bit of a binary number. Can also mean the smallest increment of change.

**logical variable** A single data bit in those cases where a single bit is used to represent an on-off switch, motor on-off control, and so on.

**local area network (LAN)** A system that allows multiple units to communicate with each other, all sharing the same interconnection wire.

**LSB** *See* **least significant bit**.

**machine language** The set of operation codes that a CPU can execute.

**memory** The part of the computer that stores digital data. Memory data is stored as bytes, where each byte is given an address.

**memory-mapped input/output** A system where I/O ports are treated exactly like memory locations.

**microcontroller** An integrated circuit that includes a microprocessor, memory, and input/output; in essence, a “computer on a chip.”

**microprocessor** A digital integrated circuit that performs the basic operations of a computer but requires some support integrated circuits to be functional.

**most significant bit (MSB)** The leftmost bit in a binary number.

**mnemonic** An English-like abbreviation of an operation code.

**modem** A circuit that converts serial data from digital form into tones that can be sent through the telephone system.

**MSB** *See* **most significant bit**.

**nonvolatile memory** Computer memory such as ROM that will not lose its data when the power is turned off.

**null modem** A cable that allows two DTE units to communicate with each other (*see* **RS-232**).

**operation code (op-code)** A digital code word used by the microprocessor to identify a particular instruction.

**parallel interface** A type of data interface where 8 bits enter or leave a unit at the same time on eight wires.



**PC** See **personal computer**.

**personal computer (PC)** A microprocessor-based, self-contained, general-purpose computer (usually refers to an IBM or compatible computer).

**PLC** See **programmable logic controller**.

**port** The part of a computer where I/O data lines are connected; each port has an address.

**program counter** A special address-holding register in a computer that holds the address of the next instruction to be executed.

**programmable logic controller (PLC)** A rugged, self-contained microprocessor-based controller designed specifically to be used in an industrial environment.

**RAM** See **random-access memory**.

**random-access memory (RAM)** Sometimes called read/ write memory, a memory arrangement using addresses where data can be written in or read out; RAM loses its contents when the power is turned off.

**read-only memory (ROM)** Similar to RAM in that it is addressable memory, but it comes preprogrammed and cannot be written into; also, it does not lose its data when the power is turned off.

**read/write (R/W) line** A control signal that goes from the microprocessor to memory.

**real time** Refers to a computer that is processing data *at the same time* that the data are generated by the system.

**resolution** In digital-to-analog conversion, the error that occurs because digital data can only have certain discrete values.

**ROM** See **read-only memory**.

**RS-232 standard** A serial data transmission standard that specifies voltage levels and signal protocol between a DTE (computer) and a DCE (modem or other device).

**R/W** See **read/write line**.

**sampling rate** The times per second a digital controller reads the sensor data.

**scan** See **iteration**.

**serial interface** A type of interface where data are transferred 1 bit after the other on a single wire.

**single-board computer** A premade microprocessor-based computer assembled onto a single printed-circuit card.

**time-delay loop** A programming technique where the computer is given a “do-nothing” job such as counting to some large number for the purpose of delaying time.

**UART** See **universal asynchronous receiver transmitter**.

**universal asynchronous receiver transmitter (UART)** A special purpose integrated circuit that converts data from parallel to serial format and vice versa.

**volatile memory** Computer memory such as RAM that will lose its data when the power is turned off.

**word** A unit of digital data that a particular computer uses; common word sizes are 4, 8, 16, and 32 bits.

## EXERCISES

### Section 2.1

1. Briefly describe the functions of the ALU, control unit, CPU memory, and input/output.
2. What steps does the microprocessor take to read data at address 1020? (Specify the actions of the address bus and data bus in your answer.)
3. Briefly define *address bus*, *data bus* and *control bus*.
4. Use the method shown in Example 2.1 to find the decimal value of the binary number 01011101.
5. Use the method shown in Example 2.1 to find the decimal value of the binary number 11011010.

### Section 2.2

6. What is a microprocessor *instruction set*, and how is it different from a high-level language such as BASIC?
7. A certain microprocessor has a simple instruction set shown below.

#### INSTRUCTION SET

<u>Op-code</u>	<u>Explanation</u>
76	Halt the microprocessor.
C6*	Add next byte to accumulator.
D6*	Subtract next byte from accumulator.
3C	Increment the accumulator.
3D	Decrement the accumulator.
3E*	Move the next byte into the accumulator.

\*These instructions use two bytes.

What number would be in the accumulator after the program shown below was run?

PROGRAM

<u>Address</u>	<u>Op-code</u>
001	3E
002	05
003	D6
004	02
005	3C
006	76

### Section 2.3

8. Temperature values from  $-20^{\circ}\text{F}$  to  $120^{\circ}\text{F}$  are input data for a microprocessor computer. Are 8 bits sufficient? If so, what is the resolution?
9. Explain the function of the following: *parallel data port* and *serial data port*.
10. Serial data are sent at 1200 bps using the format of Figure 2.11, with one stop bit. How long would it take to send 1000 bytes of data?
11. An 8-bit DAC has a reference voltage of 9 V. The binary input is 11001100. Find the analog output voltage.
12. The binary data from the computer in a certain application are expected to go from 00000000 to only 00111111. These data are the input of a DAC. The analog output should go 0-5 V. Find the DAC reference voltage necessary to make this happen.
13. An 8-bit ADC has a reference voltage of 12 V and an analog input of 3.7 V. Find the binary output.
14. The binary output of an ADC should have the range 00000000-11111111 corresponding to an input of 0-6 V. Find the necessary reference voltage.

### Section 2.4

15. What is *real-time computing*, and is it necessary for control systems?
16. Describe the basic steps in a control program scan (loop).
17. At some point in the program it is desired to have the computer wait 5 s for an operator response. How would this delay be accomplished in software?
18. A program contains 150 instructions, and the average execution time per instruction is  $2\ \mu\text{s}$ . Find the sample rate of this program.

### Section 2.5

19. What is a *microcontroller*, and what are some differences between a microcontroller and a microprocessor?
20. What is a *programmable logic controller*?
21. You want to use a personal computer to control a simple robot arm. The arm has two joints, an elbow and a wrist. Each joint has a DC motor and a position sensor that outputs a DC voltage. You already have a “plain vanilla” PC; make a list of what you would need to acquire to make this system work.
22. Compare and contrast the following: a *microprocessor*, a *microcontroller*, a *programmable logic controller*, and a *personal computer*.



# CHAPTER 3

## Operational Amplifiers and Signal Conditioning

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### OBJECTIVES

After studying this chapter, you should be able to:

- Recognize the characteristics of an operational amplifier and describe how they can be used as the basis for different types of useful amplifiers.
- Design the following types of op-amp circuits to meet specific requirements: voltage follower, inverting amplifier, noninverting amplifier, summing amplifier, differential amplifier, and comparator.
- Understand the operation of the following types of circuits: integrators and differentiators, active filters, current-loop signal transmission, analog switches and multiplexers, and sample and hold.
- Understand the concepts of the earth ground and ground loops, magnetic and electrostatic shielding, and the importance of a single-point ground.

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### INTRODUCTION

One of the necessary conditions of any real system is the successful **interfacing**, or connecting together, of the various components. In a block diagram, an interface is represented by a line between two blocks, indicating that some sort of signal passes between the blocks. If it were only this easy! The fact is, interfacing is sometimes the most difficult task in getting a system operational. There are several different categories of interfacing requirements.

One type of interfacing is between analog and digital circuits. Most controllers are digital, whereas many sensors and actuators use analog signals. This means analog-to-digital or digital-to-analog converters may be required (as discussed in Chapter 2). When designing this type of interface, you must consider such things as resolution (number of bits), analog voltage level, and conversion speed required.

Another interface problem is matching voltage levels between components. A sensor may put out a voltage range of 0-0.5 V, whereas the receiving component may need a signal in the voltage range of 0-10 V. Or a sensor may put out a high-impedance signal (easily loaded down) and needs to be converted to a stronger low-impedance signal. Still another problem is that some sensors are nonlinear, which means that the sensor output voltage is not directly proportional to the parameter being measured. Nonlinear sensors may require some correction circuitry.

Sometimes several sensors must share the same input port of a controller. This requires an electronic switching circuit capable of connecting different analog channels to the same destination. Another requirement may be to add or subtract analog signals—for example, when the feedback signal is subtracted from the set-point in an analog controller.

Some situations require that there be little or no signal loss between components, even if they are some distance apart. This might seem to be impossible because all wires have resistance; however, the *current-loop* technique virtually eliminates signal attenuation.

Another set of interface problems deals with handling electrical noise from the outside world. Although some types of interference can be filtered out, it is usually best to try to prevent the noise from entering the system. This is done with proper shielding and grounding.

We will deal with these topics in this chapter. It will not be an exhaustive treatment because whole books are available on each subject, but it will introduce accepted solutions to various problems.

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## 3.1 OPERATIONAL AMPLIFIERS

### Introduction

An **operational amplifier (op-amp)** is a high-gain linear amplifier. Op-amps are usually packaged in IC form (one to four op-amps per IC) and are relatively inexpensive. The op-amp approaches the ideal amplifier of the analog designer's dreams because it has such ideal characteristics:

1. Very high open-loop gain:  $A = 100,000+$ , but unpredictable
2. Very high input resistance:  $R_{in} > 1 \text{ M}\Omega$
3. Low output resistance:  $R_{out} = 50\text{-}75 \text{ ohm}$

These characteristics make designing with op-amps relatively easy. As we will see, the high open-loop gain makes it possible to create an amplifier with a very predictable stable gain of anywhere from 1 to 1000 or more. The significance of the very high input resistance ( $R_{in}$ ) is that the op-amp draws very little input current. This means it will not load down whatever circuit or sensor is driving it. The op-amp's low output resistance ( $R_{out}$ ) means it can drive a load without being loaded down itself. However, an op-amp is a signal amplifier, not a power amplifier. It is not designed to output large currents and so is not usually used to drive loads such as loudspeakers or motors directly.

**Figure 3.1**

The op-amp symbol.

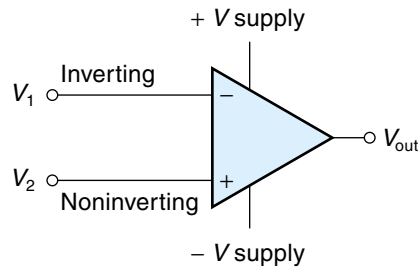


Figure 3.1 shows the symbol for a typical op-amp. It has two inputs ( $V_1$  and  $V_2$ ) and one output ( $V_{out}$ ). Also shown are the two power-supply inputs, which are typically +12 V and -12 V. The output voltage can swing to within about 80% of the supply voltages. Notice there is no ground connection at all. Most op-amps are actually differential amplifiers, which means they amplify the difference between  $V_1$  and  $V_2$ . This is shown in Equation 3.1:

$$V_{out} = A(V_2 - V_1) \quad (3.1)$$

where

$V_{out}$  = output voltage

$A$  = open-loop gain

$V_1$  = inverting input

$V_2$  = noninverting input

The **open-loop gain** ( $A$ ) is the raw unmodified gain of the op-amp; it is high, typically 100,000 or more.  $V_2$  is called the **noninverting input**. As the name implies, *the output is in phase with the noninverting input* (when the noninverting input goes positive,  $V_{out}$  goes positive; when the noninverting input goes negative,  $V_{out}$  goes negative). The noninverting input is identified by the + sign in the symbol of Figure 3.1.

The other input to the op-amp is called the **inverting input**. *The output will be out of phase with the signal at the inverting input* (when the inverting input goes more positive, the output will go more negative, and vice versa). The inverting input is identified by the - sign in the symbol.

Even though the op-amp has two separate inputs, there is just one input voltage, which is the difference between  $V_2$  and  $V_1$ . This is illustrated in Example 3.1.

### EXAMPLE 3.1

Figure 3.2 shows an op-amp with an open-loop gain of 100,000. Find the output for the following conditions:

- $V_1$  and  $V_2$  are both 4  $\mu\text{V}$ .
- $V_1$  is 2  $\mu\text{V}$ , and  $V_2$  is 4  $\mu\text{V}$ .
- $V_1$  is 6  $\mu\text{V}$ , and  $V_2$  is 3  $\mu\text{V}$ .

**SOLUTION**

We will use Equation 3.1 to solve this problem.

- a. Both  $V_1$  and  $V_2$  are  $4 \mu\text{V}$ :

$$\begin{aligned} V_{\text{out}} &= 100,000 \times (4 \mu\text{V} - 4 \mu\text{V}) \\ &= 100,000 \times (0 \mu\text{V}) \\ &= 0 \text{ V} \end{aligned}$$

This shows that the output of the op-amp is zero if the inputs are the same voltage, regardless of their actual value.

- b. The noninverting input  $V_2$  is  $4 \mu\text{V}$ , and the inverting input  $V_1$  is  $2 \mu\text{V}$ :

$$\begin{aligned} V_{\text{out}} &= 100,000 \times (4 \mu\text{V} - 2 \mu\text{V}) \\ &= 100,000 \times (2 \mu\text{V}) \\ &= 0.2 \text{ V} \end{aligned}$$

This result shows that the output is positive if the  $(V_2 - V_1)$  quantity has a net positive value.

- c. The inverting input is  $6 \mu\text{V}$ , and the noninverting input is  $3 \mu\text{V}$ :

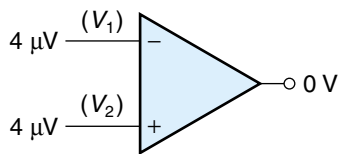
$$\begin{aligned} V_{\text{out}} &= 100,000 \times (3 \mu\text{V} - 6 \mu\text{V}) \\ &= 100,000 \times (-3 \mu\text{V}) \\ &= -0.3 \text{ V} \end{aligned}$$

This result is negative because the  $(V_2 - V_1)$  quantity has a negative net value.

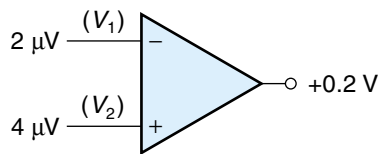
Note: The purpose of this example is only to show how an open-loop op-amp behaves. It would be very difficult to duplicate this in the lab because of the challenge of creating small, steady input voltages.

**Figure 3.2**

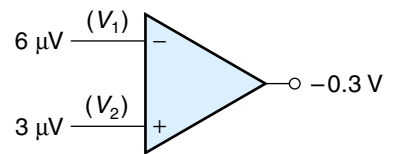
Various input voltage combinations ( $A = 100,000$ ) (Example 3.1).



(a)



(b)

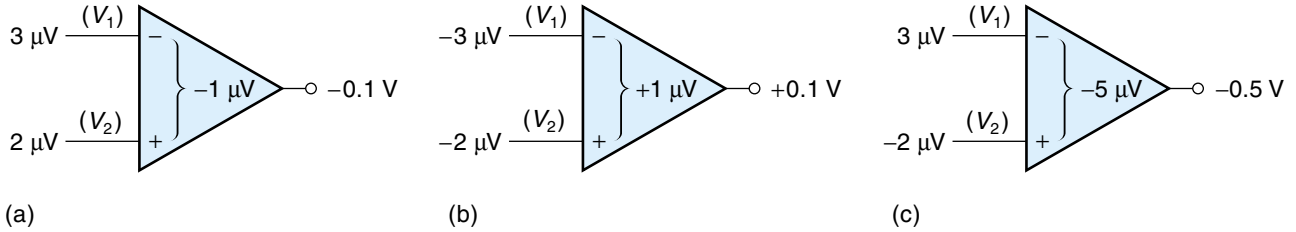


(c)



**Figure 3.3**

Various input voltage combinations ( $A = 100,000$ ).



Example 3.1 begins to illustrate one aspect of op-amps that may seem strange at first—the sign of the output. Consider the three op-amps in Figure 3.3. In Figure 3.3(a), both inputs are positive, yet the output is negative. Why? The  $-$  input has the larger magnitude, so the quantity  $(V_2 - V_1)$  is negative ( $2 \mu\text{V} - 3 \mu\text{V} = -1 \mu\text{V}$ ). From Equation 3.1 (the op-amp equation),

$$V_{\text{out}} = A(V_2 - V_1)$$

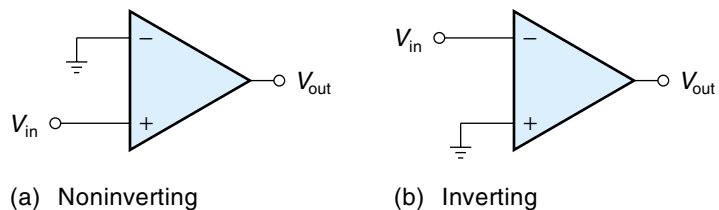
you can see that if  $(V_2 - V_1)$  is negative,  $V_{\text{out}}$  will be negative.\*

Now consider the circuit of Figure 3.3(b). The inputs are both negative, yet the output is positive. To understand this, again examine the  $(V_2 - V_1)$  quantity, paying attention to the signs. In this case,  $[-2 \mu\text{V} - (-3 \mu\text{V})] = +1 \mu\text{V}$ , which is positive. The circuit in Figure 3.3(c) is more straightforward. In this case,  $(V_2 - V_1) = (-2 \mu\text{V} - 3 \mu\text{V}) = -5 \mu\text{V}$ , which is clearly negative.

Now consider the case where only a single input is required. There are two possibilities: The output will be either in phase or out of phase with the input. To make a noninverting amplifier (where the output is in phase with the input), the  $-$  input is grounded, and the input signal is connected to the noninverting input ( $+$ ), as shown in Figure 3.4(a). If we want an inverting amplifier, where the output is out of phase with the input, we connect the signal into the inverting input ( $-$ ) and ground the  $+$  input, as shown in Figure 3.4(b).

**Figure 3.4**

Single-input, open-loop amplifiers.



\*Another way to determine the output polarity is to use the following rule: The output will assume the polarity that matches the symbol of the most positive input. In the case of Figure 3.3(a), the  $-$  input has the largest positive value, so the output is negative.

All the amplifier circuits discussed so far are called *open loop* because they operate at open-loop gain. As we will see, this is not the typical application of an op-amp—we are doing it here because it simplifies the discussion of how the differential inputs work.

Most op-amp circuits incorporate **negative feedback**. This means that a portion of the output signal is fed back and subtracted from the input. Negative feedback results in a very stable and predictable operation at the expense of lowered gain (which we can easily afford because the open-loop gain is so high to begin with). Analyzing op-amp circuits is actually easier than analyzing traditional discrete transistor amplifiers because the op-amp's impressive parameters allow us to make three circuit-simplifying assumptions:

- Assumption 1:  $V_1 = V_2$ . *Explanation:* How can we possibly assume that the inputs  $V_1$  and  $V_2$  are always the same? Could we not force  $V_1$  and  $V_2$  to be anything we like? Yes, but the argument goes like this: The output voltage is equal to  $A(V_2 - V_1)$  where  $A$ , being the open-loop gain, is a very high number. Thus, even a small difference between  $V_1$  and  $V_2$  will cause a very large output. However, the output has a practical upper limit established by the power supply; therefore, to keep the output from exceeding its limits, the difference between  $V_1$  and  $V_2$  must be very small. This is illustrated in Figure 3.5. The power supply is +15 V and -15 V, which limits the output voltage swing to about +12 V and -12 V (being 80% of the supply). If the open-loop gain is 100,000, then the difference between  $V_1$  and  $V_2$  that would cause an output of 12 V is computed using Equation 3.1:  $V_{\text{out}} = A(V_2 - V_1)$ . Rearranging gives us

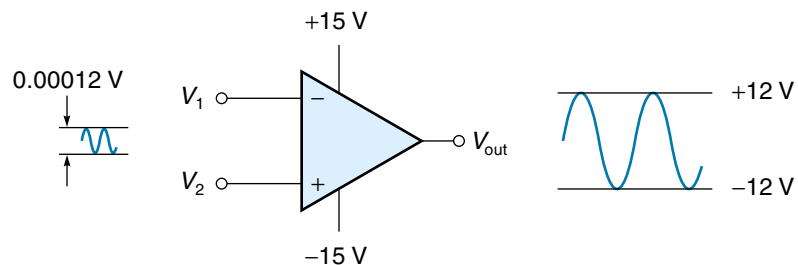
$$(V_2 - V_1) = \frac{V_{\text{out}}}{A} = \frac{12 \text{ V}}{100,000} = 0.00012 \text{ V}$$

So we see that, to keep the amplifier operating linearly with the output within its bounds, the difference between  $V_2$  and  $V_1$  must be less than 0.00012 V, which is a very small voltage. Hence, we say that  $V_1$  is *virtually the same as*  $V_2$ .

- Assumption 2: Input current is zero. *Explanation:* The input resistance of an op-amp is very high, typically 1 M $\Omega$  or more. It is so high that we can model the inputs as being open circuits as shown in Figure 3.6; of course, no current can flow into an open circuit.

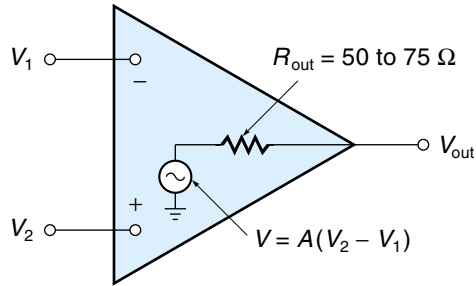
**Figure 3.5**

Op-amp inputs are always virtually the same voltage.



**Figure 3.6**

Equivalent circuit model of an op-amp.



- Assumption 3: Output resistance is zero. *Explanation:* A low-output resistance means that the output voltage will not be pulled down even if the load draws a lot of current. This is the weakest of the three assumptions because the output resistance is typically between 50 and 75  $\Omega$  (however, it can be much lower with negative feedback). This assumption only holds if the load being driven is considerably higher than the output resistance of the op-amp, which is the case in most applications.

Many different types of op-amps are available, with names such as *general-purpose*, *wide-bandwidth*, *low-noise*, and *high-frequency*, to name a few. For most control applications, the general-purpose types are adequate. Figure 3.7 shows a data sheet for the popular, general-purpose op-amp 741 (MC1741), which comes in four types of packages. Besides the inverting and noninverting input and output pins, this op-amp has two more pins called *offset null*. As indicated in the small diagram of Figure 3.7, these can be used to adjust the output voltage up or down slightly for the purpose of eliminating the **DC offset voltage**—a small DC voltage that may occur at the output, even when the inputs are exactly equal.

Looking at the electrical characteristics (Figure 3.7), the large signal voltage gain is given as 50–200 V/mV. This means that, at a minimum, the ratio is 50 V out for each millivolt in, which is the equivalent of 50,000 V out for each volt in, or a gain of 50,000 (minimum). Notice also that the input resistance is given as being typically 2 M $\Omega$  and the output resistance is typically 75  $\Omega$ .

The 741 is a traditional op-amp that is commonly cited in textbooks because people are familiar with it and it is usually stocked in labs. However, there are hundreds of types of op-amps on the market, many of them newer and better in some way than the 741. Fortunately, nearly all of them have the same pin configuration as the 741, so you could design a circuit around a 741 and then substitute a different model if you needed the performance. Examples of newer op-amps include the LF355, LM308, and LF411. The LF411 has an extremely high input resistance and practically no offset voltage.

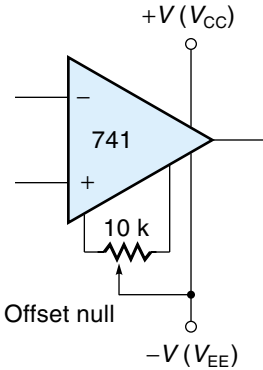
Many useful signal-conditioning circuits can be built using op-amps. Some of the most common are presented in the pages that follow.

## Voltage Follower

The **voltage follower**, which is very useful circuit, can boost the current of a signal without increasing the voltage. It can transform a high-impedance signal (easily loaded

**Figure 3.7**

Data sheet for the 741 general-purpose op-amp. (Copyright © Semiconductor Components Industries, LLC. Used by permission.)



**MC1741**  
**INTERNALLY COMPENSATED, HIGH PERFORMANCE OPERATIONAL AMPLIFIERS**

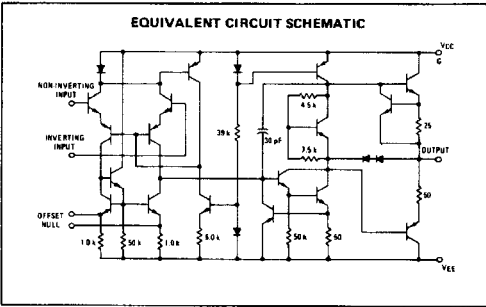
... designed for use as a summing amplifier, integrator, or amplifier with operating characteristics as a function of the external feedback components.

- No Frequency Compensation Required
- Short-Circuit Protection
- Offset Voltage Null Capability
- Wide Common-Mode and Differential Voltage Ranges
- Low-Power Consumption
- No Latch Up
- Low Noise Selections Offered – N Suffix

**MAXIMUM RATINGS** (T<sub>A</sub> = +25°C unless otherwise noted)

Rating	Symbol	MC1741C	MC1741	Unit
Power Supply Voltage	V <sub>CC</sub> V <sub>EE</sub>	+18 -18	+22 -22	Vdc Vdc
Input Differential Voltage	V <sub>ID</sub>	±30		Volts
Input Common Mode Voltage (Note 1)	V <sub>ICM</sub>	±15		Volts
Output Short Circuit Duration (Note 2)	t <sub>SC</sub>	Continuous		
Operating Ambient Temperature Range	T <sub>A</sub>	0 to +70	-55 to +125	°C
Storage Temperature Range	T <sub>stg</sub>	-65 to +150	-55 to +125	°C
Junction Temperature Range	T <sub>J</sub>	175	150	°C

Note 1: For supply voltages less than ±15 V, the absolute maximum input voltage is equal to the supply voltage.  
 Note 2: Supply voltage equal to or less than 15 V



**OPERATIONAL AMPLIFIER SILICON MONOLITHIC INTEGRATED CIRCUIT**

**G SUFFIX METAL PACKAGE CASE 601**

**P1 SUFFIX PLASTIC PACKAGE CASE 626 (MC1741C, MC1741NC)**

**U SUFFIX CERAMIC PACKAGE CASE 693**

**L SUFFIX CERAMIC PACKAGE CASE 632 TO-116**

**P2 SUFFIX PLASTIC PACKAGE CASE 646 (MC1741C, MC1741NC)**

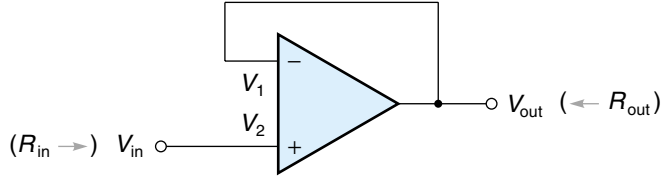
**F SUFFIX CERAMIC PACKAGE CASE 606-04 TO-91**

**ELECTRICAL CHARACTERISTICS** (V<sub>CC</sub> = +15 V, V<sub>EE</sub> = -15 V, T<sub>A</sub> = 25°C unless otherwise noted).

Characteristic	Symbol	MC1741			MC1741C			Unit
		Min	Typ	Max	Min	Typ	Max	
Input Offset Voltage (R <sub>G</sub> ≤ 10 k)	V <sub>IO</sub>	—	1.0	5.0	—	2.0	6.0	mV
Input Offset Current	I <sub>IO</sub>	—	20	200	—	20	200	nA
Input Bias Current	I <sub>IB</sub>	—	80	500	—	80	500	nA
Input Resistance	r <sub>i</sub>	0.3	2.0	—	0.3	2.0	—	MΩ
Input Capacitance	C <sub>i</sub>	—	1.4	—	—	1.4	—	pF
Offset Voltage Adjustment Range	V <sub>IOR</sub>	—	±15	—	—	±15	—	mV
Common Mode Input Voltage Range	V <sub>ICR</sub>	±12	±13	—	±12	±13	—	V
Large Signal Voltage Gain (V <sub>O</sub> = ±10 V, R <sub>L</sub> ≥ 2.0 k)	A <sub>v</sub>	50	200	—	20	200	—	V/mV
Output Resistance	r <sub>o</sub>	—	75	—	—	75	—	Ω
Common Mode Rejection Ratio (R <sub>S</sub> ≤ 10 k)	CMRR	70	90	—	70	90	—	dB
Supply Voltage Rejection Ratio (R <sub>S</sub> ≤ 10 k)	PSRR	—	30	150	—	30	150	μV/V
Output Voltage Swing (R <sub>L</sub> ≥ 10 k) (R <sub>L</sub> ≥ 2 k)	V <sub>O</sub>	±12 ±10	±14 ±13	—	±12 ±10	±14 ±13	—	V
Output Short-Circuit Current	I <sub>OS</sub>	—	20	—	—	20	—	mA
Supply Current	I <sub>D</sub>	—	1.7	2.8	—	1.7	2.8	mA
Power Consumption	P <sub>C</sub>	—	50	85	—	50	85	mW
<b>Transient Response (Unity Gain – Non-Inverting)</b>								
(V <sub>i</sub> = 20 mV, R <sub>L</sub> ≥ 2 k, C <sub>L</sub> ≤ 100 pF) Rise Time	t <sub>TLH</sub>	—	0.3	—	—	0.3	—	μs
(V <sub>i</sub> = 20 mV, R <sub>L</sub> ≥ 2 k, C <sub>L</sub> ≤ 100 pF) Overshoot	os	—	15	—	—	15	—	%
(V <sub>i</sub> = 10 V, R <sub>L</sub> ≥ 2 k, C <sub>L</sub> ≤ 100 pF) Slew Rate	SR	—	0.5	—	—	0.5	—	V/μs

**Figure 3.8**

A voltage follower circuit.



down) into a robust low-impedance signal. Figure 3.8 shows a voltage follower circuit. It has a voltage gain of 1, with a high  $R_{in}$  and a low  $R_{out}$ . Its operation can be explained as follows: We start with the basic op-amp equation (Equation 3.1):

$$V_{out} = A(V_2 - V_1)$$

In the circuit,  $V_{out}$  is connected to  $V_1$ ; thus,  $V_{out} = V_1$ . Substituting in  $V_{out}$  and expanding Equation 3.1,

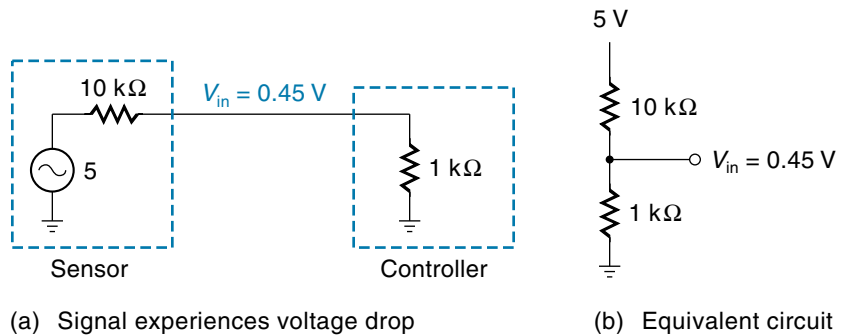
$$V_{out} = (AV_2) - (AV_{out})$$

Solving for  $V_{out}$ , we get

$$V_{out} = \frac{AV_2}{1 + A} \approx V_2$$

**Figure 3.9**

Using a voltage follower to prevent load down.



(a) Signal experiences voltage drop

(b) Equivalent circuit

(c) No signal voltage drop

But because  $A$  is *much* greater than 1,

$$V_{\text{out}} = \frac{AV_2}{A} \approx V_2$$

We see that the output voltage  $V_{\text{out}}$  equals the input voltage  $V_2$ , meaning the overall gain is 1. Also notice that the actual input signal goes directly into the noninverting input, so it draws essentially no current.

A more intuitive way to explain the voltage follower circuit is as follows: The input  $V_2$  is virtually the same voltage as  $V_1$  (from Assumption 1).  $V_1$  is connected to  $V_{\text{out}}$ , so it's as if  $V_2$  were connected to  $V_{\text{out}}$ , hence the gain of 1.

The voltage follower is a simple and very useful circuit. Consider the situation shown in Figure 3.9(a). In this case, a high-impedance sensor ( $10\text{ k}\Omega$ ), is connected directly to a controller with a  $1\text{ k}\Omega$  input resistance. The sensor generates  $5\text{ V}$  internally, but this is reduced by the voltage drop across the  $10\text{ k}\Omega$  internal resistance. By redrawing the circuit [Figure 3.9(b)], we see that these two resistances form a voltage divider. The actual input voltage to the controller can be calculated as follows from the voltage-divider rule:

$$V_{\text{in}} = \frac{1\text{ k}\Omega \times 5\text{ V}}{1\text{ k}\Omega + 10\text{ k}\Omega} = 0.45\text{ V}$$

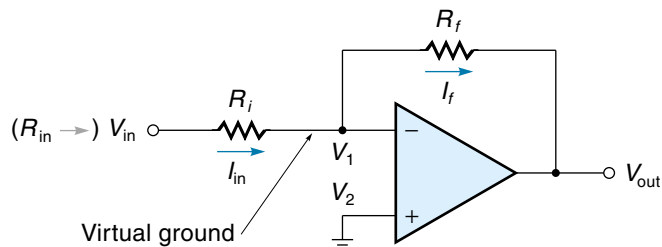
This shows that only  $0.45\text{ V}$  of the  $5\text{ V}$  signal makes it to the controller. We could amplify the signal at the controller to make up for the attenuation, but that would amplify noise as well as the signal. A better solution is to insert a voltage follower near the sensor, as shown in Figure 3.9(c). Because the op-amp draws no signal current, there is no voltage drop across the  $10\text{ k}\Omega$  resistor, and the full  $5\text{ V}$  enters the voltage follower and appears at its output. The  $1\text{ k}\Omega$  input resistance of the controller is so much higher than the output resistance of the op-amp that almost all of the  $5\text{ V}$  will appear across the controller terminals.

## Inverting Amplifier

The **inverting amplifier** is probably the most common op-amp configuration. The circuit shown in Figure 3.10 requires just two resistors,  $R_i$  and  $R_f$ .  $R_i$  is the input resistor, and  $R_f$  is the feedback resistor that feeds part of the output signal back to the input. This

**Figure 3.10**

The inverting amplifier circuit.



is an inverting amplifier because *the input signal goes to the inverting input, which means the output is out of phase with the input*. The voltage gain is determined by the resistor values.

An explanation of how the inverting amp works is as follows: First, if the op-amp input draws no current, then all the signal current ( $I_{in}$ ) must go through  $R_f$ —there is nowhere else for it to go. Therefore,  $I_{in} = I_f$ . By assumption,  $V_1$  and  $V_2$  are virtually the same voltage, and  $V_2$  is grounded; thus,  $V_1$  is at **virtual ground**. If  $V_1$  is (almost) at ground, then the entire input signal voltage  $V_{in}$  is dropped across  $R_i$ . From Ohm's law,

$$I_{in} = \frac{V_{in}}{R_i}$$

As already noted, virtually all  $I_{in}$  goes through the feedback resistor  $R_f$ . The voltage across  $R_f$  is the difference between virtual ground and  $V_{out}$ . Thus, we can write Ohm's law equation for  $R_f$ :

$$I_{in} = I_f = \frac{0 - V_{out}}{R_f}$$

Combining the two previous equations,

$$\frac{V_{in}}{R_i} = \frac{0 - V_{out}}{R_f}$$

Solving for  $V_{out}$  and rearranging gives us

$$V_{out} = \frac{-V_{in}R_f}{R_i}$$

$$\frac{V_{out}}{V_{in}} = \frac{-R_f}{R_i}$$

However,  $V_{out}/V_{in}$  is the voltage gain, so

$$A_v = \frac{-R_f}{R_i} \quad (3.2)$$

where

$A_v$  = voltage gain of the inverting amp

$R_f$  = value of the feedback resistor

$R_i$  = value of the input resistor

This result (Equation 3.2) shows us that the voltage gain of the inverting amp is simply the ratio of the two resistors  $R_f$  and  $R_i$ . The minus sign reminds us that the output is inverted. The gain derived in Equation 3.2 is called the **closed-loop gain** and is always lower than the (open-loop) gain of the op-amp by itself.

Another important point is that the input impedance for the entire inverting amp is approximately  $R_i$  (not infinite as one might think). Figure 3.10 shows this: The right end of  $R_i$  is at virtual ground; therefore, the entire  $V_{in}$  is “dropped” across  $R_i$ .

### EXAMPLE 3.2

An inverting amp is to have a gain of 10. The signal source is a sensor with an output impedance of 1 k $\Omega$ . Draw a circuit diagram of the completed amplifier.

### SOLUTION

First, select a value for  $R_i$ . Because  $R_i$  essentially determines the amplifier’s input resistance, *it should be at least ten times higher (if possible) than the signal source impedance to ensure maximum voltage transfer*. In this example, we select  $R_i = 10$  k $\Omega$ . Next, rearrange Equation 3.2 to solve for  $R_f$ :

$$\begin{aligned} R_f &= -AR_i \\ &= -(-10) \times 10 \text{ k}\Omega = 100 \text{ k}\Omega \end{aligned}$$

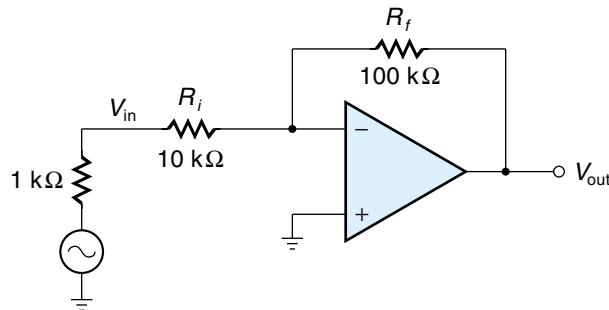
Figure 3.11 shows the completed circuit.

## Noninverting Amplifier

Many situations call for an amplifier that does not invert the output. For example, the output of a temperature sensor might be such that as the temperature goes up, the voltage goes up. If this is the same relationship that the controller wants, we don’t want the amplifier to invert it. The circuit for the **noninverting amplifier** is shown in Figure 3.12. It is similar to the inverting amp except the input signal  $V_{in}$  now goes directly to the noninverting input and  $R_i$  is grounded. Notice that the noninverting amp has an almost infinite input impedance ( $R_i$ ) because  $V_{in}$  connects only to the op-amp input.

**Figure 3.11**

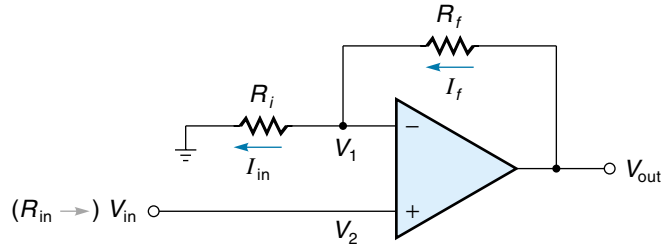
An inverting amplifier circuit (Example 3.2).





**Figure 3.12**

The noninverting amplifier circuit.



An explanation of how the circuit works is as follows: If  $V_1$  is virtually the same as  $V_2$ , then the voltage input ( $V_{in}$ ) appears across  $R_i$ . Applying Ohm's law to  $R_i$ , we can calculate  $I_{in}$ :

$$I_{in} = \frac{V_{in} - 0}{R_i}$$

The current in  $R_f$  can also be calculated using Ohm's law. We know the voltage across  $R_f$  is the difference between  $V_{in}$  and  $V_{out}$ . Therefore,

$$I_f = \frac{V_{out} - V_{in}}{R_f}$$

Because no current enters the inverting input of the op-amp, all current in  $R_f$  must go into  $R_i$ :

$$I_{in} = I_f$$

Combining these three equations gives us

$$I_{in} = I_f = \frac{V_{in} - 0}{R_i} = \frac{V_{out} - V_{in}}{R_f}$$

Solving for  $V_{out}$  and rearranging gives us

$$\begin{aligned} V_{out} - V_{in} &= \frac{R_f V_{in}}{R_i} \\ V_{out} &= \frac{R_f V_{in}}{R_i} + V_{in} = V_{in} \left( \frac{R_f}{R_i} + 1 \right) \\ \frac{V_{out}}{V_{in}} &= \frac{R_f}{R_i} + 1 \end{aligned}$$

$V_{out}/V_{in}$  is the voltage gain, so the resulting equation for the gain of the noninverting amp is

$$A_V = \frac{R_f}{R_i} + 1 \quad (3.3)$$

where

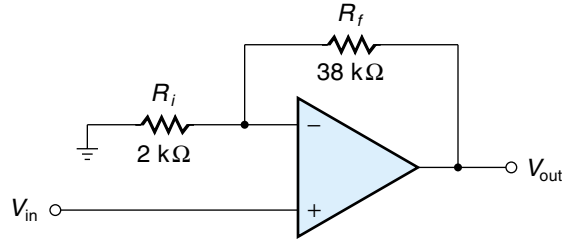
$A_V$  = voltage gain for the noninverting amp

$R_f$  = value of the feedback resistor

$R_i$  = value of the input resistor

**Figure 3.13**

A noninverting amplifier circuit (Example 3.3).

**EXAMPLE 3.3**

Draw the circuit diagram of a noninverting amp with a gain of 20.

**SOLUTION**

Using Equation 3.3 and putting in a gain of 20,

$$A_V = 20 = \frac{R_f}{R_i} + 1$$

Rearranging gives us

$$\frac{R_f}{R_i} = 19 \quad \text{or} \quad R_f = 19 \times R_i$$

Now select  $R_i$  to be an appropriate value (as explained below) and solve for  $R_f$ . If we select  $R_i$  to be 2 kΩ, then

$$R_f = 19 \times R_i = 19 \times 2 \text{ k}\Omega = 38 \text{ k}\Omega$$

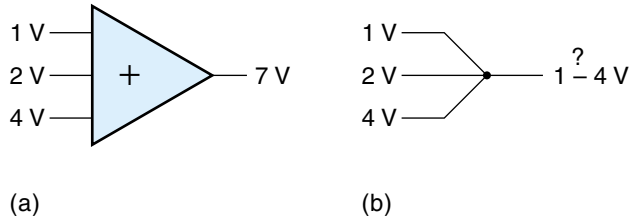
Figure 3.13 shows the completed circuit. The basis for selecting both  $R_i$  and  $R_f$  is that the current in these external resistors should be much larger than the small current that actually enters the op-amp (recall that the op-amp equation was based on the assumption that *no* current enters the op-amp). Therefore, both  $R_i$  and  $R_f$  should be at least ten times smaller than the op-amp input resistance—in this case, no more than 100 kΩ if possible.

**Summing Amplifier**

The **summing amplifier** has an output voltage that is the sum of any number of input voltages. Figure 3.14(a) depicts this situation. In this case, the amplifier would add the input voltages of 1 V, 2 V, and 4 V and give an output of 7 V. You might be tempted to

**Figure 3.14**

Connecting the wires does not sum the voltage.



think you could do this by simply connecting the wires as shown in Figure 3.14(b), but the output of that circuit would be something between 1 and 4 V depending on which source was the “strongest” (had the lowest resistance).

Figure 3.15 shows the summing amplifier circuit. Each input signal goes through a separate resistor that keeps it isolated from the others. The inverting input is used, so the output will be inverted. An explanation of how it works is as follows: Because the op-amp input draws no current, all individual input currents must combine and go through the feedback resistor  $R_f$ :

$$I_f = I_a + I_b + I_c \tag{3.4}$$

Note that  $V_2$  is grounded, so  $V_1$  is at virtual ground. Therefore, the voltage across each of the four resistors is simply  $V_a$ ,  $V_b$ ,  $V_c$ , and  $V_{out}$ . Applying Ohm’s law to express the current through each resistor, we can rewrite Equation 3.4 as follows:

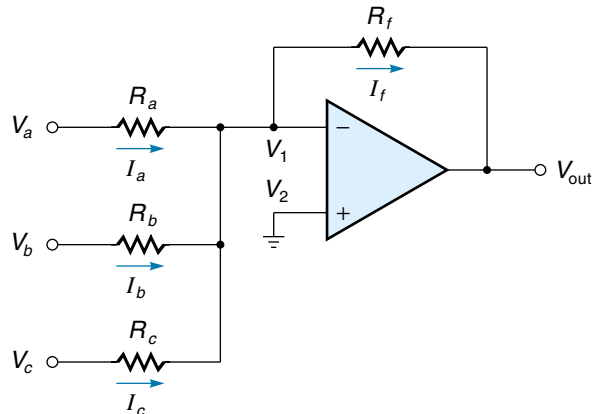
$$\frac{0 - V_{out}}{R_f} = \frac{V_a}{R_a} + \frac{V_b}{R_b} + \frac{V_c}{R_c}$$

Solving for  $V_{out}$ ,

$$V_{out} = -\left(\frac{R_f}{R_a}V_a + \frac{R_f}{R_b}V_b + \frac{R_f}{R_c}V_c\right)$$

**Figure 3.15**

The summing amplifier circuit.



In other words, the output voltage equals the sum of the products of the input voltages and their respective gains. Also, the input impedance of each input equals the value of the respective input resistor ( $R_a$ ,  $R_b$ , etc.).

If  $R_a = R_b = R_c = R_i$ , the output of the summing amp simplifies to:

$$V_{\text{out}} = -\frac{R_f}{R_i}(V_a + V_b + V_c) \quad (3.5)$$

where

$V_{\text{out}}$  = output voltage of the summing amp

$R_f$  = value of the feedback resistor

$R_i$  = value of all input resistors

$V_a$ ,  $V_b$ ,  $V_c$  = input signal voltages

Equation 3.5 shows that the output voltage  $V_{\text{out}}$  is equal to the sum of the input voltages times a gain factor of  $R_f/R_i$ . The minus sign reminds us that the output is inverted.

#### EXAMPLE 3.4

According to a comfort scale, the air conditioning in a building should come on when the sum of the temperature and humidity sensor voltages goes above 1 V. A threshold circuit in the air conditioner requires 5 V for turn-on. Design an interface circuit to connect the two sensors to the air conditioning unit.

#### SOLUTION

This circuit requires a summing amplifier with two inputs and a gain of 5. By specifying both input resistors to be the same (at 1 k $\Omega$ ), we can use Equation 3.5, and our only calculation concerns the gain portion of the equation:

$$A = \frac{R_f}{R_i} = 5$$

Rearranging gives us

$$R_f = 5 \times R_i$$

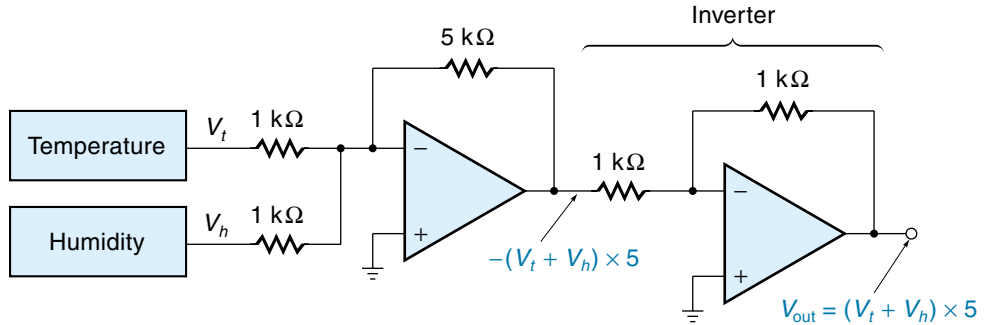
When  $R_i = 1 \text{ k}\Omega$ ,

$$R_f = 5 \times 1 \text{ k}\Omega = 5 \text{ k}\Omega$$

Figure 3.16 shows the completed circuit. Notice that an inverting amp with a gain of 1 ( $R_i = R_f$ ) was added to make the final output positive.

**Figure 3.16**

A summing amplifier circuit (Example 3.4).



### Differential and Instrumentation Amplifiers

A **differential amplifier** amplifies the difference between two input voltages. In the circuits we have examined thus far, the input voltages have been referenced to ground, but the op-amp can be the basis of a practical differential amp as well. Figure 3.17 shows such a circuit. As before, the gain is established with resistors. This circuit will amplify a **differential voltage**, which is the difference between the two voltage levels  $V_a$  and  $V_b$ , when neither is ground. The output of the amplifier ( $V_{out}$ ) is a single voltage level referenced to ground, sometimes called a **single-ended voltage**. If  $R_a = R_b$  and  $R_f = R_g$ , which is usually the case, then the equation for  $V_{out}$  is

$$V_{out} = \frac{R_f}{R_a} (V_b - V_a) \tag{3.6}$$

Rearranging gives us

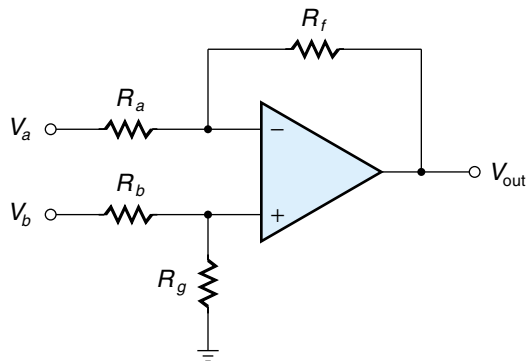
$$\frac{V_{out}}{(V_b - V_a)} = \frac{R_f}{R_a}$$

$V_{out}/(V_b - V_a)$  is output/input, which is a gain, so

$$A_V = \frac{R_f}{R_a} \tag{3.7}$$

**Figure 3.17**

The differential amplifier circuit.



where

$A_V$  = voltage gain of the differential amp

$R_f$  = value of  $R_f$  and  $R_g$

$R_a$  = value of input resistors,  $R_a$  and  $R_b$

As with the basic op-amp, the polarity of the output ( $V_{\text{out}}$ ) will be positive when the input  $V_b$  is more positive than input  $V_a$ . The selection of resistor values  $R_a$  and  $R_b$  is based on a compromise. If they are too high—say, over 100 k $\Omega$ —then the currents may be so small that our basic op-amp assumptions won't hold (for example, the assumption that all current in  $R_a$  goes through  $R_f$ ). On the other hand, if the resistances are too low (less than ten times the source resistance), then a considerable amount of attenuation will occur before the signal even gets to the amp.

### EXAMPLE 3.5

A differential amp is needed to amplify the voltage difference between two temperature sensors. The sensors have an internal resistance of 5 k $\Omega$ , and the maximum voltage difference between the sensors will be 2 V. Design the differential amp circuit to have an output of 12 V when the difference the inputs is 2 V.

### SOLUTION

First calculate the gain required:

$$A_V = \frac{V_{\text{out}}}{V_{\text{in}}} = \frac{12\text{V}}{2\text{V}} = 6$$

By letting  $R_a = R_b$  and  $R_f = R_g$ , we can use Equation 3.7. Noting that the sensor impedance is 5 k $\Omega$ , we would like the input resistance of the amp to be at least ten times 5 k $\Omega$ . Therefore, if we select  $R_a = 50$  k $\Omega$ , then

$$A_V = \frac{R_f}{R_a} = 6$$

solving for  $R_f$

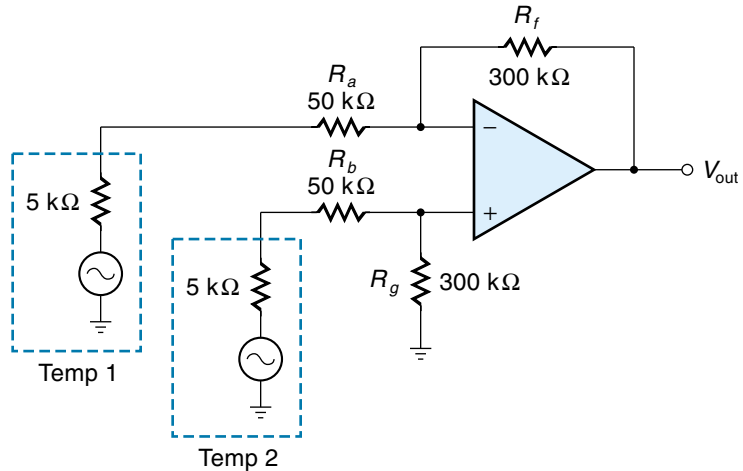
$$\begin{aligned} R_f &= R_a \times 6 \\ &= 50 \text{ k}\Omega \times 6 = 300 \text{ k}\Omega \end{aligned}$$

Figure 3.18 shows the completed design.

Sometimes information is transmitted as a *differential signal*, which means that the actual signal voltage is the difference between two wires (where neither wire is ground). Such a signal could be converted into a regular (ground-referenced) voltage by a dif-

**Figure 3.18**

A differential amplifier circuit (Example 3.5).



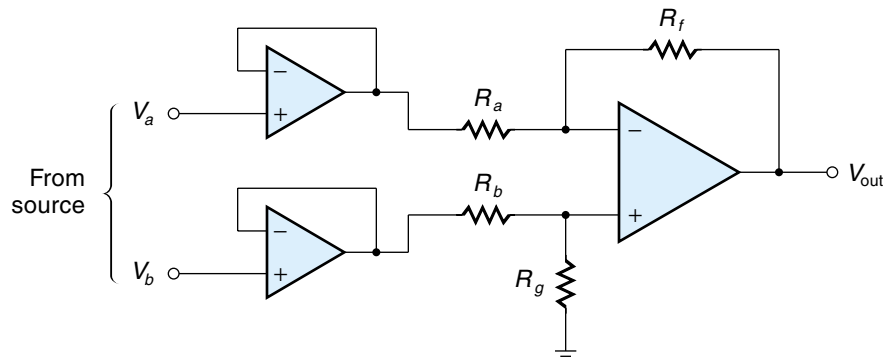
ferential amplifier. The advantage of this system is that it reduces the effect of electrical noise. A noise signal would tend to couple onto both wires; for example, a positive voltage noise “spike” would cause a positive spike in *both* wires, which would be canceled out by the differential amplifier (because only voltage differences are amplified). This is called **common mode rejection**.

An **instrumentation amplifier** is a differential amp that has its inputs buffered with voltage followers, as shown in Figure 3.19. Voltage follower circuits on the inputs perform three desirable functions: (1) They increase the input resistance so that the source (such as a sensor) will never be loaded down, (2) they make both input resistances equal, and (3) they isolate the gain-defining resistors ( $R_p$ ,  $R_i$ , etc.) from the signal source. This last quality means that instrumentation amps can be prebuilt to have a specific gain.

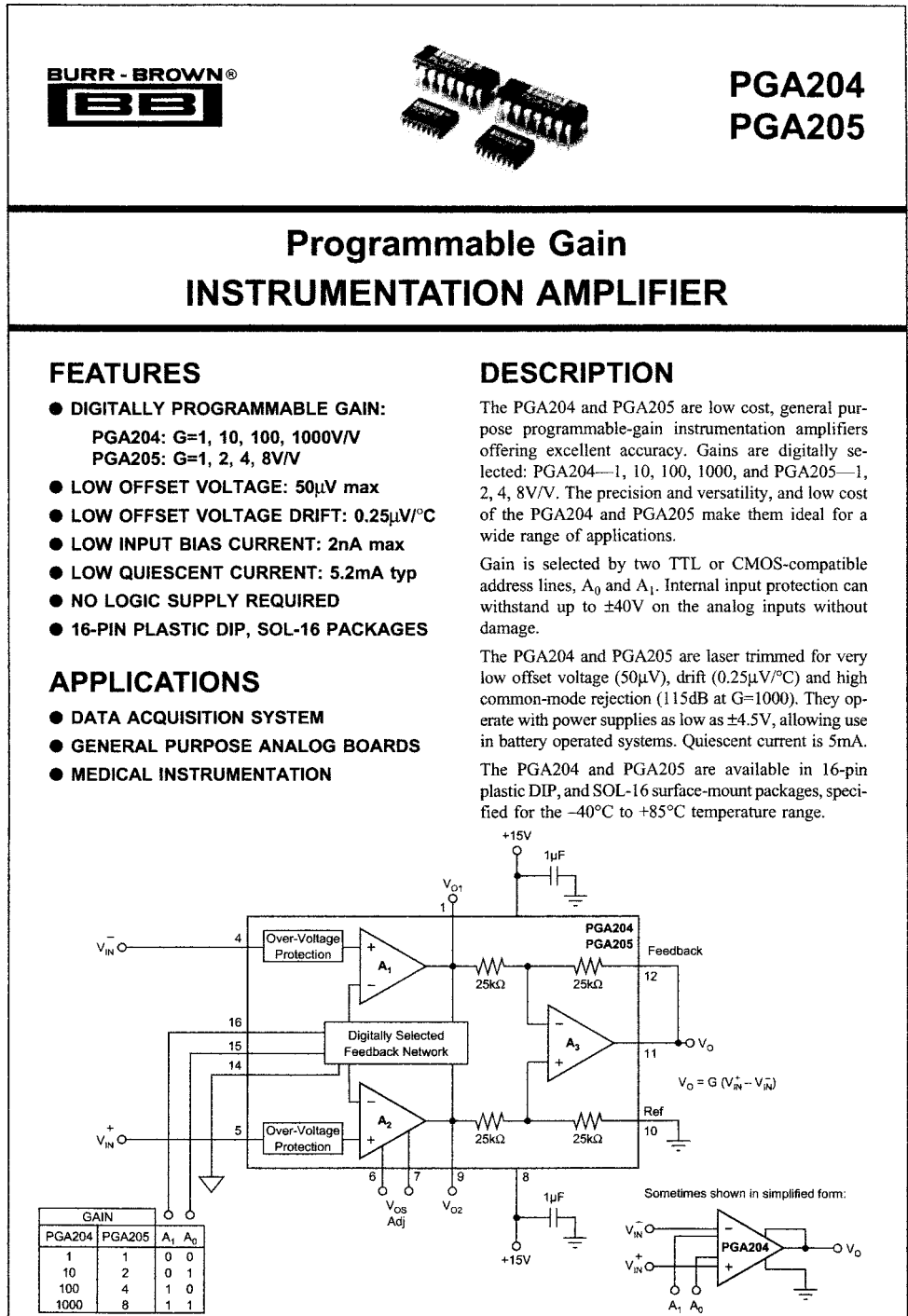
A data sheet for the Burr-Brown PGA204 instrumentation amplifier is shown in Figure 3.20. This type of instrumentation amplifier is known as a **programmable gain instrumentation amplifier** because the gain is selectable with digital inputs. Packaged

**Figure 3.19**

The instrumentation amplifier circuit.



**Figure 3.20**  
Data sheet for the PGA204 instrumentation amplifier. (Courtesy of Burr-Brown Corp.)





as a 16-pin IC, the PGA204 has four fixed gains of 1, 10, 100, and 1000, which are selected by two digital inputs (00, 01, 10, 11). These amplifiers are useful in applications where signals of vastly different voltages are digitized by the same ADC (analog-to-digital converter). In such a system, the digital controller could switch to the appropriate gain for each different signal level.

## Integrators and Differentiators

Op-amp circuits can be designed to integrate or differentiate an incoming waveform. These special-purpose circuits are likely to be found only inside an analog controller (as discussed in Chapter 11).

Figure 3.21 shows an **integrator** circuit. Notice that the feedback element is a capacitor. The integrator gives an output voltage ( $V_{\text{out}}$ ) that is proportional to the total area under a curve traced out by the input voltage waveform (the horizontal axis being time), as specified in Equation 3.8:

$$V_{\text{out}} = -\frac{1}{RC} \times (\text{area under } V_{\text{in}} \cdot \text{time curve}) \quad (3.8)$$

where

$V_{\text{out}}$  = output voltage of the integrator

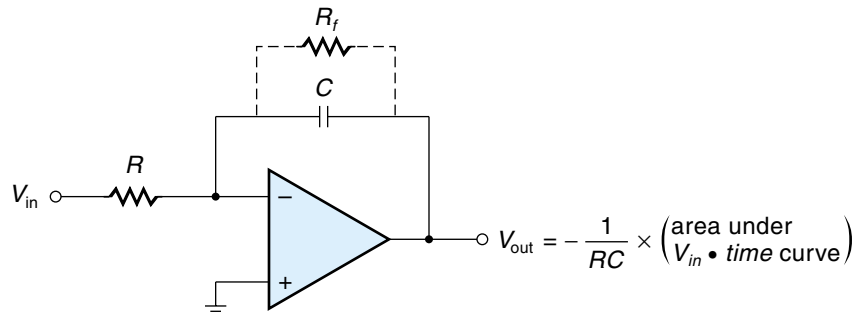
$R, C$  = values of components in Figure 3.21

The integrator circuit works by converting  $V_{\text{in}}$  into a constant current source that forces the capacitor ( $C$ ) to charge at a linear rate, thus building up the voltage across  $C$ .

The integrator concept can be explained by the sample waveforms shown in Figure 3.22 (which assumes  $RC = 1$ ). Notice that the integrator input voltage ( $V_{\text{in}}$ ) rises from 0 to 1 V in the first 10 s. The triangular area under that portion of the curve ( $a$ - $b$ ) is  $5 \text{ V} \cdot \text{s}$ , so the output ( $V_{\text{out}}$ ) of the integrator goes from 0 to  $-5 \text{ V}$  during the same time. In other words, the output voltage ends up being the same magnitude as the area under the curve, in this case 5 (the minus sign appears because it is an inverting amp). From time  $b$  to  $c$ ,  $V_{\text{in}}$  remains at 1 V, so the new area added is  $10 \text{ V} \cdot \text{s}$ . Consequently, the magnitude of  $V_{\text{out}}$  increases by 10 to become  $-15 \text{ V}$  at time  $c$ . Then,  $V_{\text{in}}$  returns to 0 V. Because no new

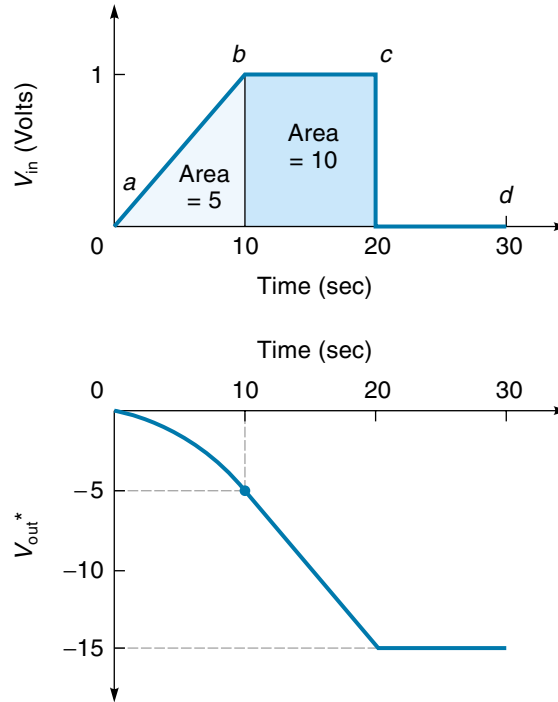
**Figure 3.21**

An integrator circuit.



**Figure 3.22**

The voltage waveform of an integrator circuit. (\* $V_{out} = -V \cdot s$  in this case because  $RC = 1$ .)



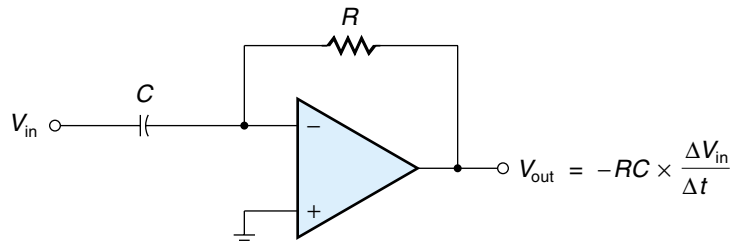
area is added between  $c$  and  $d$ ,  $V_{out}$  remains at  $-15$  V. Integrators can be useful because they keep a record of what has gone on before.

The simple integrator circuit shown is not practical, because any offset voltage (and there is always some) will eventually cause the output to build up and saturate at the power supply voltage. One solution is to put a resistor ( $R_f$ ) across the capacitor to provide some dc feedback. If the value of  $R_f$  is at least 10 times greater than  $R$ , the circuit performance will usually not be adversely affected.

Figure 3.23 shows a **differentiator** circuit. The differentiator gives an output voltage that is proportional to the rate of change (slope) of the input voltage, as specified in Equation 3.9:

**Figure 3.23**

A differentiator circuit.



$$V_{\text{out}} = -\frac{1}{RC} \times \frac{\Delta V_{\text{in}}}{\Delta t} \quad (3.9)$$

where

$V_{\text{out}}$  = output voltage of the differentiator

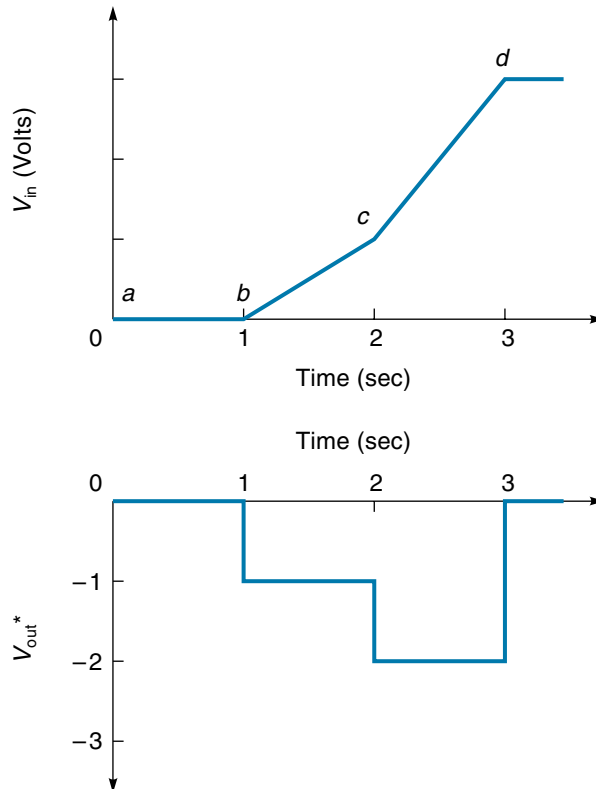
$R, C$  = values of components in Figure 3.23

$\Delta V_{\text{in}}/\Delta t$  = rate of change, or slope, of  $V_{\text{in}}$

The differentiator concept is illustrated in Figure 3.24 (which assumes  $RC = 1$ ). From time  $a$  to  $b$ , the input voltage ( $V_{\text{in}}$ ) is 0 V, and because it is not changing, the output voltage ( $V_{\text{out}}$ ) is 0 V. During the time period  $b$ - $c$ ,  $V_{\text{in}}$  increases at a constant rate of 1 V/s, so the  $V_{\text{out}}$  curve reflects this by staying at a constant  $-1$  V (it is negative because the inverting input is used). From time  $c$  to  $d$ , the slope of  $V_{\text{in}}$  increases to 2 V/s, so  $V_{\text{out}}$  jumps to  $-2$  V. After time  $d$ ,  $V_{\text{in}}$  stays at 3 V, and because it is not changing,  $V_{\text{out}}$  is 0. Differentiators can tell us how fast a variable is changing. In practice, however, they suffer from the problem that even a small amount of noise in the input will be accentuated, giving a very “noisy” output.

**Figure 3.24**

Voltage waveforms of a differentiator. ( $*V_{\text{out}} = -V_{\text{in}}$  this case because  $RC = 1$ .)



**Decibels (db)**

Amplifier and filter gains are frequently expressed in decibels (db), named after Alexander Graham Bell. The **decibel** is an expression, on a logarithmic scale, of how much greater (or smaller) the output is than the input. The original idea was to use decibels to express power gains using Equation 3.10.

$$Ap_{\text{db}} = 10 \log Ap \quad (3.10)$$

where

$Ap_{\text{db}}$  = power gain in decibels

$Ap$  = simple power gain (called the “straight gain”)

As it turns out, it is more common to measure voltage gain than power gain, because voltage gains are easier to measure and calculate. The equation for decibels voltage gain is given in Equation 3.11 and used in Example 3.6 to calculate the decibels gain of an amplifier.

$$Av_{\text{db}} = 20 \log Av \quad (3.11)$$

where

$Av_{\text{db}}$  = voltage gain in decibels

$Av$  = straight voltage gain

**EXAMPLE 3.6**

An op-amp design has a voltage gain of 5000. Find the voltage gain in decibels.

**SOLUTION**

Apply Equation 3.11 to find the voltage gain in decibels.

$$Av_{\text{db}} = 20 \log Av = 20 \log 5000 = 74\text{db}$$

Therefore, the amplifier voltage gain expressed in decibels is 74 db.

**EXAMPLE 3.7**

An amplifier is said to have a voltage gain of 45 db. Find the straight voltage gain.

**SOLUTION**

Rearrange Equation 3.11 to solve for  $Av$ .

$$A_v = \text{antilog} \frac{A_{db}}{20} = 10^{(A_{db}/20)}$$

$$A_v = 10^{(45/20)} = 10^{2.25} = 178$$

Therefore, the straight voltage gain of the amplifier is 178.

The db system tends to expand small-gain numbers and compress the large-gain numbers. This can be clearly in Table 3.1. Looking at the table, we can make further observations:

1. A straight gain of 1 (meaning *no* gain or loss) is 0 db.
2. A straight gain of 2 (meaning the output is twice as great as the input) corresponds to 6 db.
3. A straight gain of 0.5 (which is really an attenuation because the output is only half as great as the input) corresponds to -6 db. You can see that all attenuations are negative decibels.
4. When a straight gain goes up (or down) by a factor of 10, the decibel gain changes by adding (or subtracting ) 20 db.

Using these observations, we can come up with a simple procedure that will allow you to come up with a close estimate of decibel gain with pencil and paper (or in your head!). This procedure is demonstrated in Example 3.8, which uses the following rules:

1. Factor the straight voltage gain into multiples of 10 and multiples of 2.
2. For each multiple of 10, *add* 20 db
3. For each multiple of 2, *add* 6 db.

**TABLE 3.1**

**Voltage Gain Expressed in Decibels**

Straight voltage gain, $A_v$	Decibel gain, $A_{v_{db}}$
0.01	-40 db
0.1	-20 db
0.25	-12 db
0.5	-6 db
1	0 db
2	6 db
4	12 db
10	20 db
100	40 db
1000	60 db
10,000	80 db

**EXAMPLE 3.8**

Estimate the decibel gain for the amplifier of Example 3.6 .

**SOLUTION**

The amplifier of Example 3.6 had a straight voltage gain of 5000. The first thing to do is factor 5000 into multiples of 2 and of 10 as best we can.

$$\begin{aligned} 5000 &= 5 \times 10 \times 10 \times 10 && \text{(Notice that 5 is almost 4, which is } 2 \times 2.) \\ &\approx 2 \times 2 \times 10 \times 10 \times 10 \end{aligned}$$

Now we add 6 db for each 2× and add 20 db for each 10×

$$6 + 6 + 20 + 20 + 20 = 72 \text{ db}$$

Notice that our estimate of 72 db is pretty close to the actual answer of 74 db.

You can reverse this procedure to estimate the straight voltage gain from the decibel gain. This is demonstrated in Example 3.9, which uses the following rules:

1. Break up the decibel gain into the closest match of addends of 6 and 20.
2. For each addend of 6, multiply by 2 (for each addend of −6, multiply by 0.5)
3. For each addend of 20, multiply by 10 (for each addend of −20, multiply by 0.1)

**EXAMPLE 3.9**

Estimate the straight voltage gain for the amplifier of Example 3.7.

**SOLUTION**

The amplifier had a voltage gain of 45 db. The first thing to do is break up 45 db into addends of 6 db and 20 db as best we can.

$$\begin{aligned} 45 \text{ db} &= 5 \text{ db} + 20 \text{ db} + 20 \text{ db} && \text{(Notice that 5 is almost 6.)} \\ &\approx 6 \text{ db} + 20 \text{ db} + 20 \text{ db} \end{aligned}$$

Now multiply by 2 for each 6 db, and multiply by 10 for each 20 db.

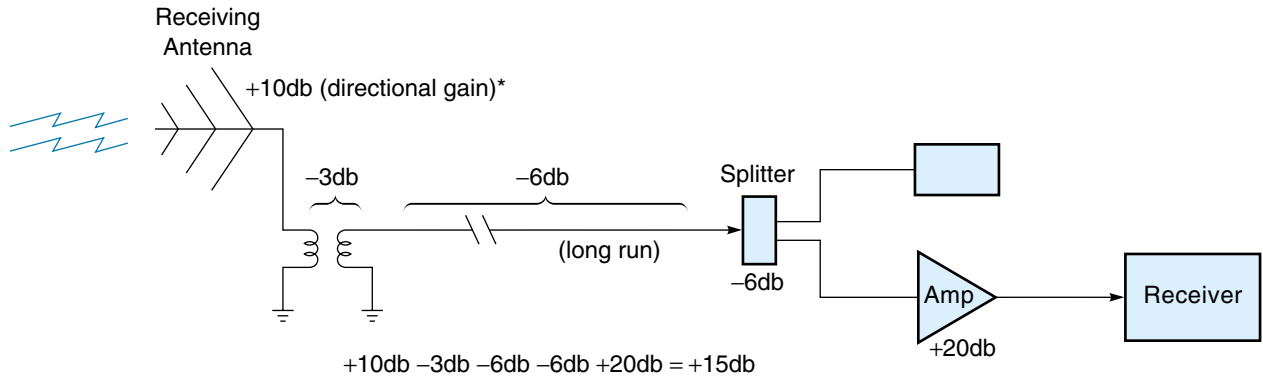
$$2 \times 10 \times 10 = 200$$

Notice that our answer of 200 is pretty close to the actual answer of 178.

Using decibels makes it easier to calculate the overall gain of a system. A signal may have to go through many modules on its way to its final destination, and each module could act as either an amplifier or an attenuator. If you know the decibel gain for each module, all you have to do is *add them up* to get the overall gain (straight gains have to

**Figure 3.25**

Total decibel gain in an antenna system.



\*The design of the antenna can boost the gain in a certain direction over what a simple antenna could do.

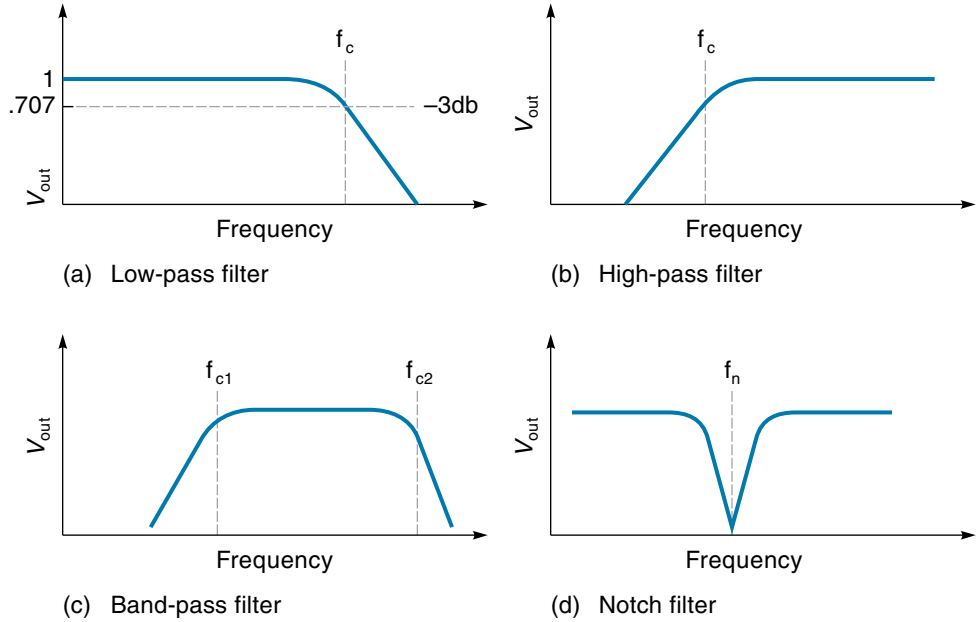
be multiplied). This principle is demonstrated in Figure 3.25 for an antenna system. You can see that the signal goes through four modules, each with its own gain or loss, but the overall gain is easily calculated to be 15 db.

### Active Filters

Filter circuits either pass or stop signals, depending on frequency. Figure 3.26 shows the responses of some basic types of filters. The **low pass filter** [Figure 3.26(a)] allows only frequencies below the **cutoff frequency** ( $f_c$ ) to pass. Frequencies above  $f_c$  are attenuated. The actual cutoff frequency is defined as the frequency where the gain drops to 0.707 (-3 db) of its pass-band value. The steepness of the attenuation depends on the type of filter. A common single-stage  $R$ - $C$  filter (described below) decreases the signal by a factor of 10 each time the frequency goes up by 10 (or -20 db per decade). The **high-pass filter** [Figure 3.26(b)] tends to reject signals with frequencies below the cutoff frequency and pass those above. The **band-pass filter** [Figure 3.26(c)] passes signals with a range of frequencies between  $f_{c1}$  and  $f_{c2}$  and rejects all others. The **notch filter** [Figure 3.26(d)] rejects only a narrow range of frequencies and passes all others.

Some of these filters are particularly useful for signal conditioning in control systems. For example, sensors reporting such things as temperature or flow rate have relatively slow-changing signals. A low-pass filter would allow the sensor signal to pass while rejecting higher-frequency electrical noise, as from motors or relays. The notch filter is also particularly useful. It can be used to attenuate a particular noise frequency, such as 60 Hz, and pass everything else.

**Figure 3.26**  
Basic filter responses.

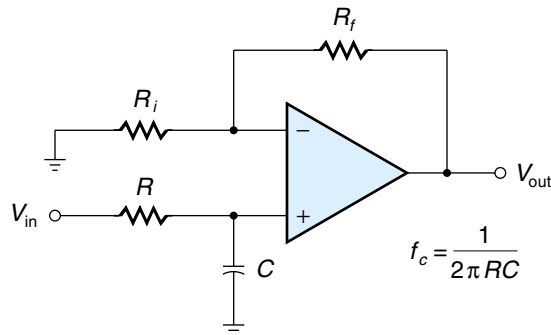


Theoretically, filters can be constructed entirely from passive components ( $R$ ,  $C$ ,  $L$ ). In practice, such filters tend to change characteristics when inserted in a circuit because the impedances interact. The op-amp can solve this problem by providing impedance isolation between the filter and the circuit it's driving, and provide some gain as well. A filter using an op-amp is called an **active filter**. Figure 3.27 shows a single-stage low-pass filter incorporating an op-amp. Notice that this is basically a noninverting amp with an  $R$ - $C$  filter connected to the input. The performance of this filter can be described with Equations 3.3 and 3.12:

$$\text{Gain} = A_V = \frac{R_f}{R_i} + 1 \quad (3.3 \text{ repeated})$$

$$\text{Cutoff frequency (low- and high-pass filters)} = f_c = \frac{1}{2\pi RC} \quad (3.12)$$

**Figure 3.27**  
A low-pass filter circuit.





**Figure 3.28**

A high-pass filter circuit.

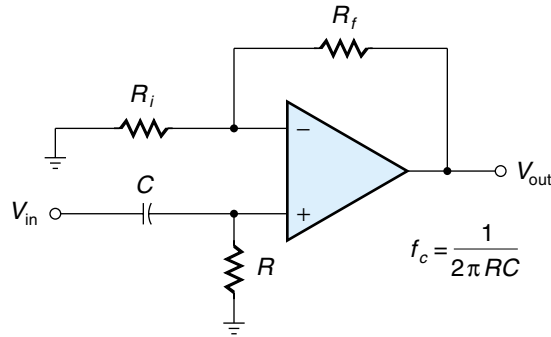


Figure 3.28 shows a single-stage high-pass filter. It is similar to the low-pass filter but the positions of the  $R$  and  $C$  are reversed. The equations for the gain and cutoff frequency are exactly the same as for the low-pass filter (but the meaning of  $f_c$  is different—see Figure 3.26).

A band-pass filter can be built by cascading a low-pass filter and a high-pass filter together. The cutoff frequency of the low-pass filter ( $f_{c2}$ ) must be higher than the cutoff frequency of the high-pass filter ( $f_{c1}$ ), as indicated in Figure 3.26(c).

Figure 3.29 shows an op-amp notch filter. The filter itself is a Wein Bridge type. The gain of the signal is 1 at all frequencies except near the notch. The frequency ( $f_n$ ), which is suppressed or “notched out,” can be calculated from Equation 3.13:

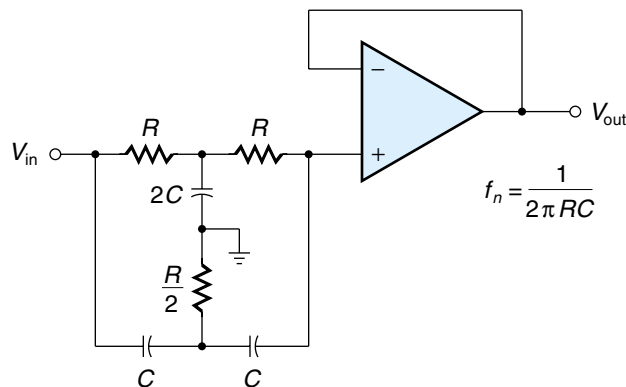
$$\text{Notched frequency (Wein Bridge notch filter)} = f_n = \frac{1}{2\pi RC} \quad (3.13)$$

### Comparator

A typical situation in a control system is a slow-moving analog signal from a sensor being used to trigger some event. Such an interface requires a threshold detector circuit

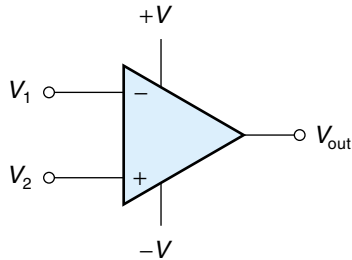
**Figure 3.29**

A Wein Bridge notch filter circuit.



**Figure 3.30**

A comparator circuit.



that will switch from off to on when a specified input voltage level is reached. A **comparator** is such a circuit (Figure 3.30). It is really an op-amp that is specially designed for this application (regular op-amps usually aren't stable enough). Comparators are usually operated open-loop so that if  $V_2$  is even slightly more positive than  $V_1$ , the tremendous gain will amplify the small difference and drive the output into positive saturation (close to  $+V$ ). On the other hand, if  $V_1$  is slightly more positive than  $V_2$ , the output will go to negative saturation ( $-V$ ). The output is essentially digital in nature—either on or off depending on a very small change in the inputs. This concept is best demonstrated in an example.

### EXAMPLE 3.10

The blower on a hot-air solar panel should come on when the temperature reaches  $100^\circ\text{F}$ . An analog temperature sensor in the solar panel needs to be interfaced to a digital controller such that the controller receives a 5 V switch-on signal when the sensor voltage reaches 2.7 V. Design the interface circuit.

### SOLUTION

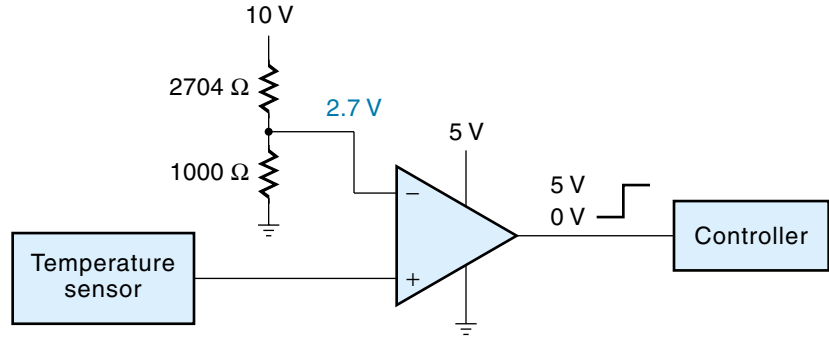
Figure 3.31(a) shows the interface circuit. The signal from the sensor is connected to the noninverting input of the comparator. The inverting input comes from a voltage divider that yields a precise reference voltage of 2.7 V. Notice also that the supply voltages of the comparator are 5 V and ground.

As long as the sensor voltage is below 2.7 V, the reference voltage at the inverting input predominates, and the output will try to go negative. In this case, the output will go to about 0 V because that is what the negative supply voltage is. When the sensor voltage goes only slightly above 2.7 V, the noninverting input becomes positive compared with the inverting input, and the output saturates positive, which is about 5 V. The switch-on point can easily be adjusted by changing the reference voltage resistors.

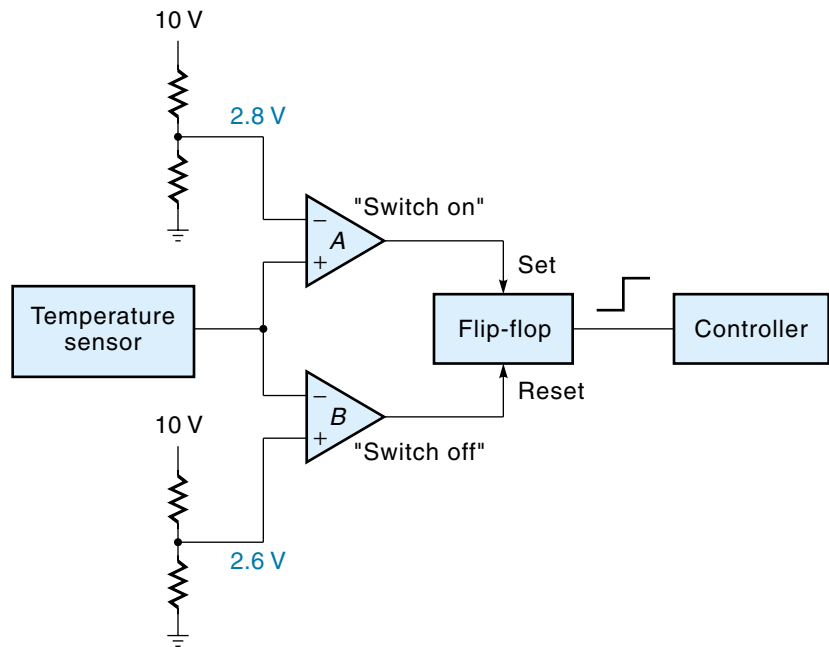
One practical problem with comparators is known as **chatter**, the condition that occurs when the output ( $V_{\text{out}}$ ) oscillates back-and-forth when  $V_{\text{in}}$  is near the threshold. Chatter is caused by noise on the  $V_{\text{in}}$  signal or some sort of undesirable feedback,

**Figure 3.31**

Comparator circuits.



(a) Comparator circuit for example 3.10

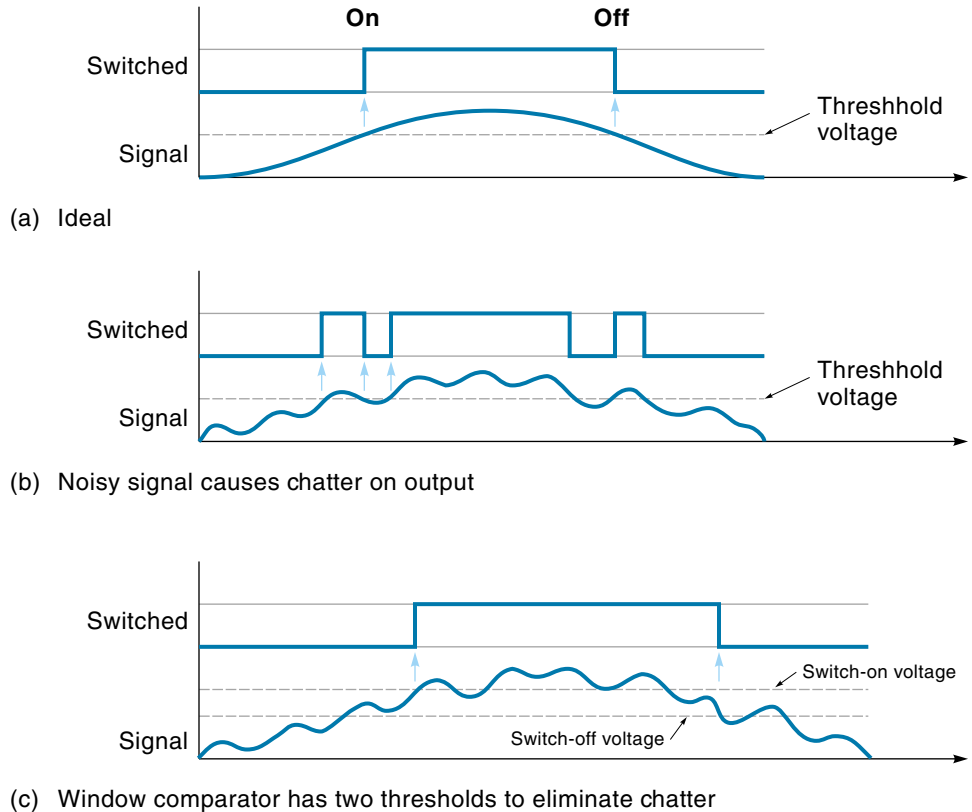


(b) Window comparator circuit

say, through the power supply. Practical circuits overcome chatter by using a **window comparator** [Figure 3.31(b)], a comparator with built-in hysteresis. In this discussion, *hysteresis* means that the switch-on voltage will be greater than the switch-off voltage. For example, if hysteresis were added to the system of Example 3.10, the switch-on voltage might be 2.8 V, and the switch-off voltage might be 2.6 V. Illustrating this situation, Figure 3.32(a) shows the ideal single-threshold system with a noiseless signal. Figure 3.32(b) shows how chatter results if the signal is noisy.

**Figure 3.32**

How a window comparator eliminates chatter.



Finally, Figure 3.32(c) shows how a system with a switch-on and switch-off threshold can eliminate chatter.

Figure 3.31(b) shows one type window comparator that is built from two op-amps and an *R-S* flip-flop. When the temperature-sensor voltage rises above 2.8 V, op-amp A switches on and sets the flip-flop. Once set, the flip-flop will stay set until the temperature voltage goes below 2.6 V, in which case op-amp B will switch on and reset the flip-flop.

## 3.2 SPECIAL INTERFACE CIRCUITS

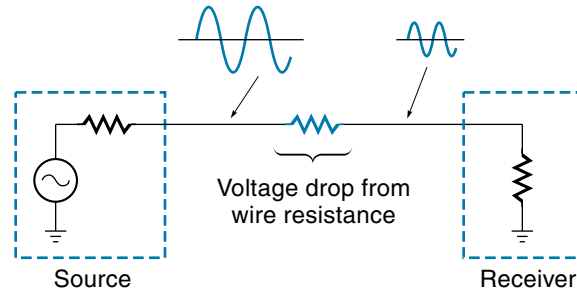
A number of special-purpose interface circuits are useful in certain applications. Some common ones are discussed in this section.

### The Current Loop (Voltage–Current Conversion)

Most signals are voltage signals, which means that the information being conveyed is proportional to the voltage. Two potential problems with this can arise: Susceptibility

**Figure 3.33**

Signal voltage is reduced from wire resistance.



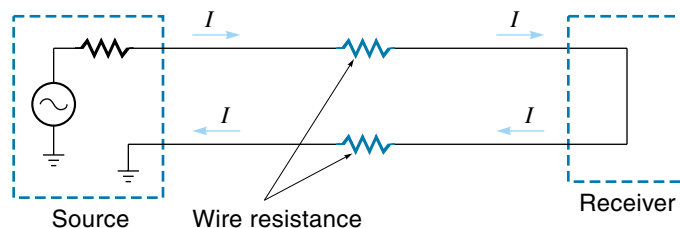
to electrical noise increases (which is usually in the form of voltage spikes), and any resistance in the signal wire causes a voltage drop. For short distances, wire can usually be effectively shielded from noise, and voltage drops are not a problem. For longer cable runs with numerous connectors, such as might be found in a process control system in a large factory, noise and total cable resistance may become significant. Figure 3.33 illustrates this situation. Notice that the signal from the sensor has been greatly diminished by the cable resistance.

One solution to the problems of noise and signal attenuation is to use current instead of voltage to convey the information. This is effective because current, unlike voltage, is not as susceptible to noise and does not drop when it goes through a resistance. As Kirchhoff's current law tells us, whatever current goes into a branch, comes out. This is illustrated in Figure 3.34. Notice that the signal emerges from the source, travels to the receiver, and then loops back again to the source. The receiver does not siphon off any current but merely senses it; therefore, we have a **current loop**, which goes from source to receiver and back to source. Any resistance in the wires cannot alter the fact that the current remains the same throughout the entire loop.

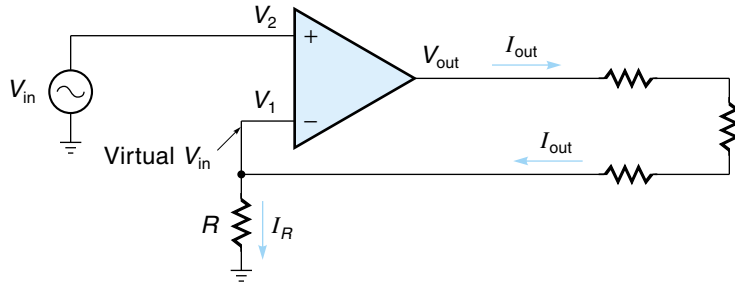
Op-amps can be used to implement both the transmitter and the receiver in a current-loop system. Figure 3.35 shows a transmitter circuit. This circuit converts a voltage signal ( $V_{in}$ ) into an output current ( $I_{out}$ ), which is proportional to  $V_{in}$ . The op-amp circuit causes the output current to be independent of any resistance in the line. It accomplishes this by automatically increasing or decreasing its output voltage ( $V_{out}$ ) in response to any increasing or decreasing line resistance.

**Figure 3.34**

A current-loop circuit.



**Figure 3.35**  
The op-amp as a current-loop transmitter.



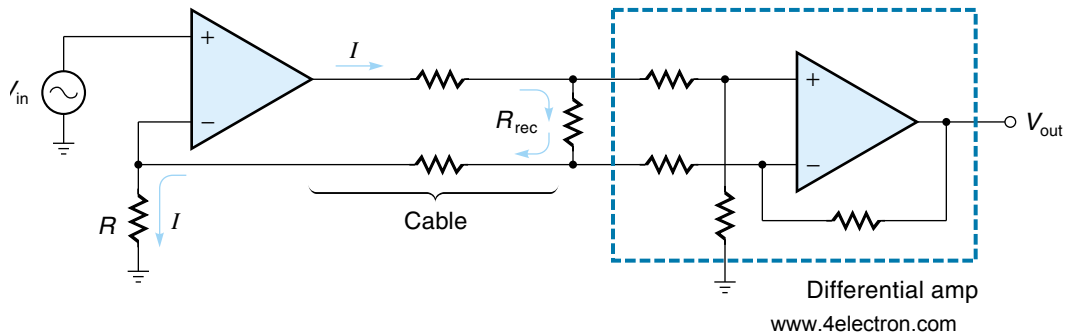
The explanation for how the circuit of Figure 3.35 works is as follows: Recall that for an op-amp we can assume  $V_1$  is virtually the same as  $V_2$ ; therefore, the voltage across  $R$  must be  $V_{in}$ . Applying Ohm's law to  $R$ , we get

$$I_R = \frac{V_{in}}{R} \quad (3.14)$$

where  $I_R$  is the value of the current in resistor  $R$ . The source of  $I_R$  cannot be the  $-$  input of the op-amp, so it must be coming through the loop. The loop current (as specified in Equation 3.14) is dependent only on the input signal voltage ( $V_{in}$ ) and the resistor  $R$ , not by any line or load resistance.

Once the current signal gets to the receiver, it needs to be reconverted back into a voltage. In other words the receiver is a **current-to-voltage converter**. A resistor is the most direct way to convert a current into a voltage, but we have a special problem here. *We have to detect the voltage across the load resistor, without knowing where this voltage is with respect to ground.* Consider the complete current-loop circuit in Figure 3.36. The receiver resistor ( $R_{rec}$ ) is “floating,” that is, neither end is grounded. (If we ground the bottom of  $R_{rec}$ , current can “escape” from the loop, and the whole concept will fall apart.) The problem is solved by using a differential amplifier as the receiver, as illustrated in Figure 3.36. The differential amplifier samples and amplifies only the voltage difference across the receiver resistor. The differential amp must be designed to have a high input resistance to prevent any significant amount of current from leaving the loop at the receiver. *The output of the differential amp is a single-ended voltage that can be referenced to ground.* To calculate this voltage:

**Figure 3.36**  
The complete current-loop system.



$$\text{Voltage across the receiver resistor: } V_{\text{rec}} = IR_{\text{rec}}$$

where

$I$  = loop current

$R_{\text{rec}}$  = receiver resistance

$$\text{Output voltage of the receiver: } V_{\text{out}} = A_v V_{\text{rec}} = A_v IR_{\text{rec}}$$

where

$A_v$  = differential amp voltage gain

The current-loop technique is one of the industry standard methods of connecting controllers to sensors and/or actuators. The standard system specifies a current range of 4-20 mA, where 4 mA corresponds to the minimum signal and 20 mA corresponds to the maximum signal. For example, if a sensor puts out a signal in the range of 0-5 V, then 4 mA in the current loop would correspond to 0 V from the sensor, and 20 mA would correspond to 5 V. The nonzero minimum (4 mA) is used to detect circuit opening and to minimize the effect of electrical noise. If the current is less than 4 mA, then the assumption is that something has failed. Transmitters and receivers in the range 4 to 20 mA are available in IC form; for example, the Burr-Brown RCV420 receiver converts 4-20 mA to 0-5 V.

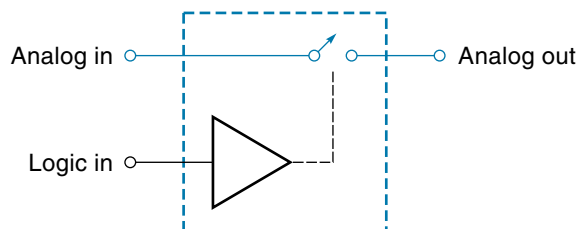
### Analog Switch Circuit

An **analog switch** (also known as a *transmission gate*) is a solid-state device that allows analog signals to pass through or not. It performs the same function as a mechanical switch but has some definite advantages: It is smaller and more reliable, it uses less power but is faster acting, and it doesn't have mechanical parts to wear out. The analog switch is usually activated by a digital control signal.

Figure 3.37 shows the symbol of an analog switch. Each switch has three terminals: analog in, analog out, and logic in. Some models allow signals to pass in either direction like a mechanical switch, whereas others are unidirectional. The logic-in input is the control signal that closes or opens the switch. Typically, the circuits are packaged in an IC. Figure 3.38 shows the ADG 201A (Analog Devices) which contains four separate analog switches.

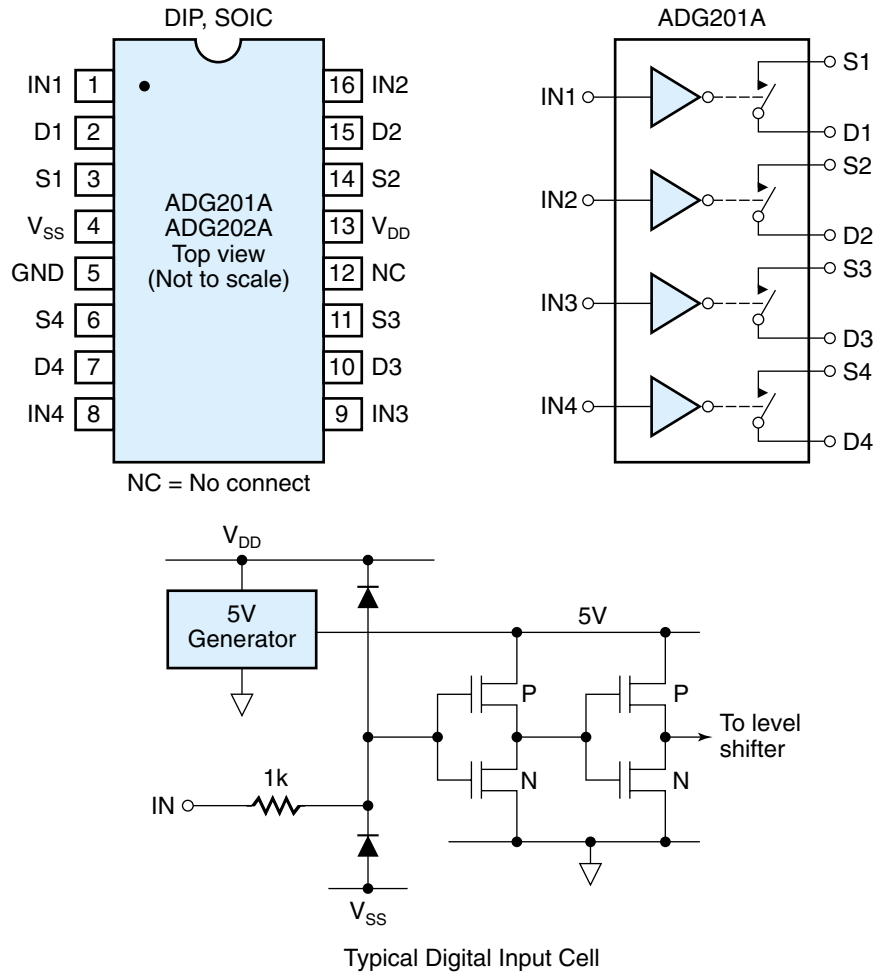
The switch itself is an FET (field effect transistor) (FETs are discussed in Chapter 4). Being a semiconductor, the FET has some resistance even when the switch is on

**Figure 3.37**  
The analog  
switch symbol.



**Figure 3.38**

The ADG 201A quad SPST analog switch circuits and schematic diagram. (Courtesy of Analog Devices)



(closed). This “on” resistance is typically 60  $\Omega$ . To keep this resistance from dropping signal voltage, the load resistance should be much greater than 60  $\Omega$ . The “off” resistance of the analog switch is very high, typically 1 M $\Omega$  or more.

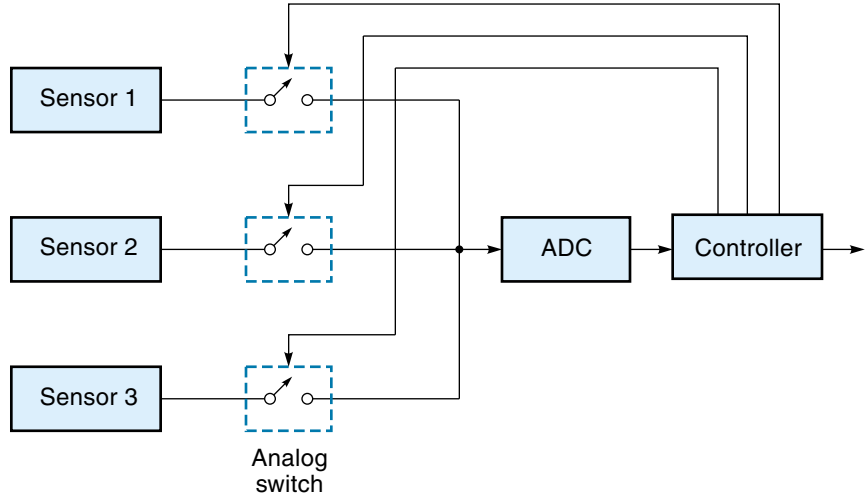
The analog switch is frequently used to connect two or more sensors to a single ADC. Figure 3.39 shows a system where three sensors are serviced by one ADC. The controller can activate each analog switch, thus connecting itself with the sensor it wants to read. The operation of switching through one input at a time is called **multiplexing**. Analog multiplexing circuits are available in IC form. Figure 3.40 shows the ADG 508A (Analog Device), an eight-channel analog multiplexer. A binary code placed on inputs A0, A1, and A2 will cause one of the eight input signals to be switched through to the output.

The ADC0809 multiplexer from National Semiconductor (not shown) is an IC that includes an 8-bit multiplexer and an ADC in one package. It is a very useful chip and found in many designs.



**Figure 3.39**

Three sensors switched into a single analog-to-digital converter (ADC).

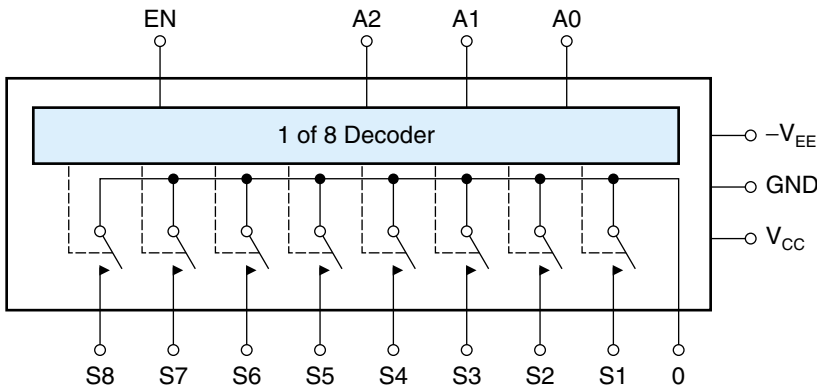


### Sample-and-Hold Circuit

A **sample-and-hold circuit** can read in, or sample, a voltage and then remember, or hold, it for a period of time. The function of this circuit can best be described in an example. Consider the system discussed in the previous section, where three analog signals were interfaced to a digital controller. Now add an additional constraint that all three sensors be read at exactly the same time. Assuming that just one ADC is available, the only way to meet this requirement is to include three sample-and-hold circuits as shown in Figure 3.41. By command from the controller, all three sample-and-hold circuits will take voltage readings and store the results. Then at the convenience of the system, these values can be read one-at-a-time by the controller.

**Figure 3.40**

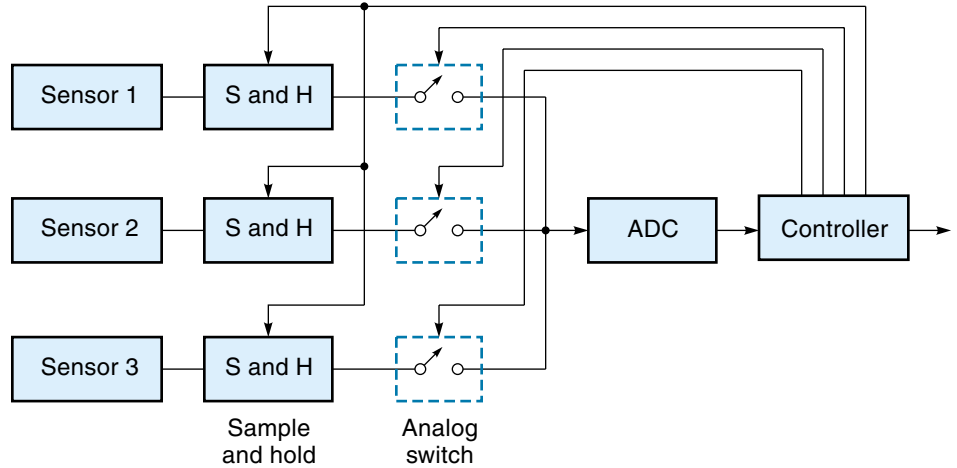
The ADG 508A eight-channel analog multiplexer circuit.



EN	A2	A1	A0	Switch On
H	L	L	L	S1
H	L	L	H	S2
H	L	H	L	S3
H	L	H	H	S4
H	H	L	L	S5
H	H	L	H	S6
H	H	H	L	S7
H	H	H	H	S8
L	X	X	X	none

**Figure 3.41**

An example of using sample-and-hold circuits.



There is another reason why a sample-and-hold circuit might be used in conjunction with an ADC. For the ADC to give an accurate output, the analog input should be constant during the read-in time. If the analog signal is changing too fast, the sample-and-hold circuit can be used to “freeze” the input voltage during the conversion.

Figure 3.42 shows the schematic of a sample-and-hold circuit, which consists of an analog switch, a capacitor, and a voltage follower amplifier. To take a sample, the analog switch is closed for a period of time long enough for the capacitor to charge up to  $V_{in}$ ; then the switch is opened. The signal voltage is trapped on the capacitor because (theoretically) it can’t discharge through either the op-amp or analog switch. The voltage can be read anytime via the output of the voltage follower. Recall that the gain of a voltage follower is 1.

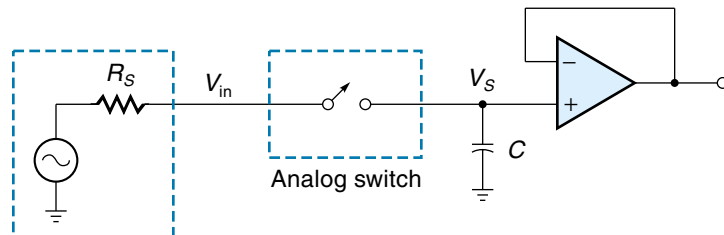
For the sample-and-hold circuit to work, certain conditions must be met. First, the switch must be closed long enough for the capacitor to charge up to the full value of  $V_{in}$ . The time required depends on the sizes of  $R_s$  (source resistance) and  $C$ , according to Equation 3.15 (which is based on  $5 RC$  time constants):

$$t = 5R_sC \tag{3.15}$$

where  $t$  is the time for  $C$  to charge to 99% of  $V_{in}$ . Eventually the charge will leak off, so the controller must read the capacitor voltage within a certain time. Herein is one of

**Figure 3.42**

A sample-and-hold circuit.



the trade-offs in the circuit design. A larger capacitor will hold the charge longer, but it will increase the read-in time because it takes longer to be charged up. As always, some compromise value of  $C$  is selected.

---

### 3.3 SIGNAL TRANSMISSION

A control system that performs just fine on a laboratory test bench may not work at all in the real world. In the lab, the cable runs are short, and the electrical noise is minimum. The real installation may require cable runs of perhaps hundreds of feet, in a dirty, electrically noisy environment. In this section, we examine some of the problems and their solutions in the area of signal transmission.

#### Earth Ground and Ground Loops

**Earth ground** refers to an electrical connection to the surface of the earth, usually through a metal rod driven into the ground. Because the earth is not a particularly good conductor, the ground voltage is different from place to place. Often confused with the earth ground is the **signal common** (or **signal return**), which is a conductor in the circuit and serves as a common reference point for the other circuit voltages. The signal common of a circuit is often referred to as the *ground*, and it may or may not be connected to the earth ground. For a DC circuit, the signal common is usually connected to the negative terminal of the power supply or battery. An automobile chassis ground is an example of this. AC signals also have a designated signal return.

With the exception of antenna systems, circuits do not have to be attached to the earth to work, but most are.\* This is done primarily for safety reasons so that high voltages cannot build up between the chassis and the local ground. Consider the situation illustrated in Figure 3.43(a). A 200-Vac signal is generated in one unit and sent by two wires to a second unit. The signal common is connected to the chassis, but the chassis is not connected to the ground. This circuit would work fine electrically, but there is a potential safety hazard. Due to inductive and capacitive coupling, it is likely that the 200 V signal voltage would align itself as shown, +100 V for the signal and -100 V for the signal common (and chassis). This is a dangerous situation because anyone standing on the ground and touching the chassis would get a 100 V shock. In Figure 3.43(b), the chassis has been connected to an earth ground. Now the voltage of the chassis is 0 V with respect to the earth and is safe to touch, as is the signal common wire.

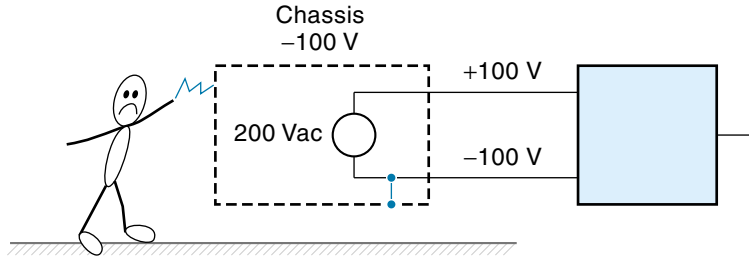
Sometimes both the source and the receiver are earth-grounded, as shown in Figure 3.44(a). One might be tempted to use the earth as the return path, but this won't work because the earth is not a good conductor and probably the two ground points are not

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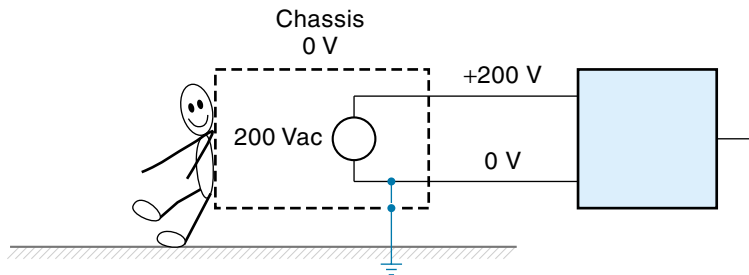
\*Mobile and other nongrounded transmitters must make do with an artificial ground plane, such as the hood of a car, or they inductively and capacitively couple to real ground.

**Figure 3.43**

Connecting the signal common to an earth ground for safety.



(a) Common connected to chassis only

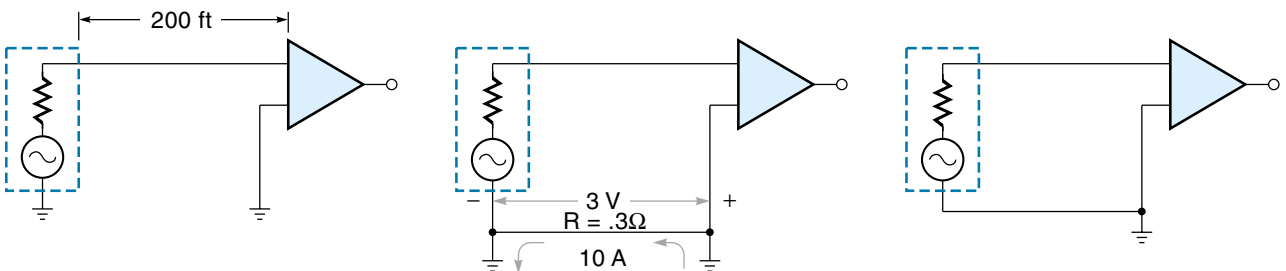


(b) Common connected to chassis and earth

at the same voltage. Assume that the cable in Figure 3.44(a) is 200 ft long and the source voltage ranges from 0 to 2 V. The difference in earth voltage for that distance could easily be 3 V (due to local conditions). It's a case where the "noise" (3 V) is larger than the signal (2 V). Connecting a separate signal return wire, as shown in Figure 3.44(b), is not a good idea because we now have a **ground loop**. A ground loop occurs when large currents flow in the return wire because of the difference in ground voltages. In the circuit of Figure 3.44(b), the wire resistance is  $0.3 \Omega$  (for wire size AWG 12), and if there really is a 3 V difference between the two ground points, the ground-loop current would be  $3 \text{ V} / 0.3 \Omega = 10 \text{ A}$ . Clearly, it is not desirable to have large, unpre-

**Figure 3.44**

Grounding signal returns can cause ground loops.

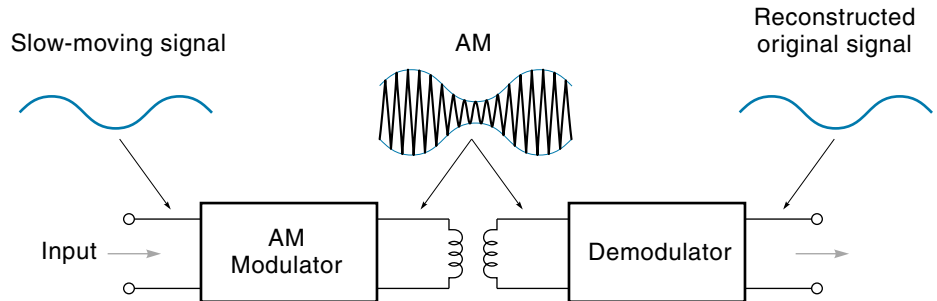


(a) Using earth as common (not good!)

(b) Connecting grounds causes a "ground loop"

(c) Grounded at one end (no "ground loop")

**Figure 3.45**  
A transformer-coupled isolation circuit.



dictable currents running around in your cabling! The solution to the ground-loop problem is indicated in Figure 3.44(c). This circuit does have a return wire for the signal, but it is grounded only at one end.

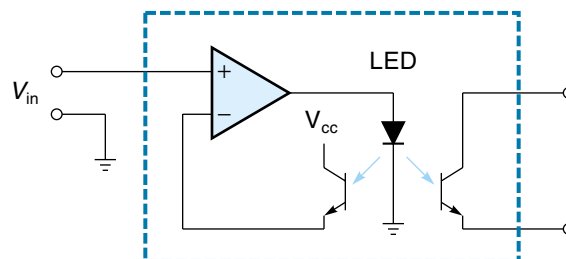
### Isolation Circuits

The ground-loop problem can be avoided by simply not connecting both ends of the signal return to an earth ground. However, sometimes the ground connections have already been made inside the equipment, in which case the only solution is to use **isolation circuits** between the source and receiver. An isolation circuit can transfer a signal voltage without any metal-to-metal contact. This allows for the signal common of the source to be at a different voltage than the signal common of the receiver.

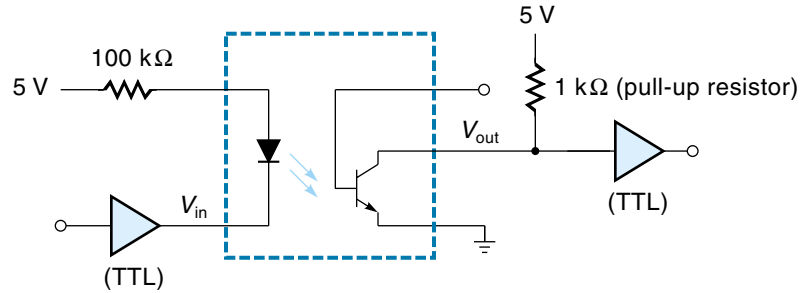
Figure 3.45 shows a transformer-coupled isolation circuit. The input voltage is used to amplitude modulate an internally generated AC voltage. The modulated AC voltage is transformer-coupled to a demodulator, which reconstructs the original input voltage. The modulation is necessary because most sensor outputs are slow-changing DC, not AC. Because there is no electrical connection across the transformer coils, the signal commons can exist at different voltages, and the ground-loop path is broken.

Figure 3.46 shows an optical linear signal isolator. The input signal is amplified to drive a light-emitting diode (LED). A phototransistor receives the light signal and converts it back to a voltage. Isolation is achieved because the input and output are two separate circuits with no electrical interconnection. To ensure that the output is an exact copy of the input, a feedback system using a second phototransistor is incorporated.

**Figure 3.46**  
An optical linear signal isolator.



**Figure 3.47**  
Optical coupler  
(for digital signals).

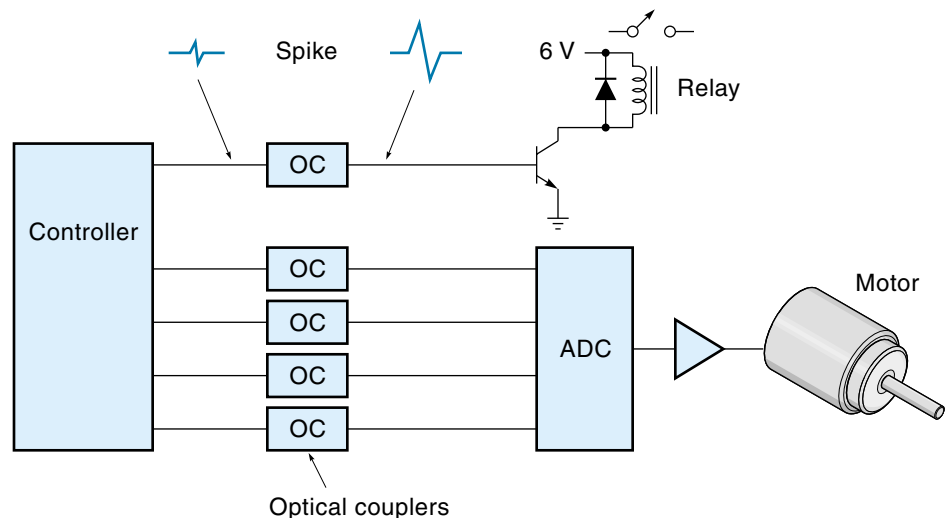


A simpler optical device can be used if the signal to be isolated is digital. Digital signals do not require a linear relationship between input and output (just on and off). Figure 3.47 shows a typical digital **optical coupler**. A standard TTL digital device can drive the input. When the input ( $V_{in}$ ) is driven low, a voltage is developed across the LED causing it to light. This light turns on the phototransistor, which pulls its output ( $V_{out}$ ) to 0 V. The output of the photocoupler can be connected directly to a TTL gate with a pull-up resistor as shown.

Optical couplers (sometimes called opto-isolators) are very useful in providing noise isolation because signals can only travel in one direction through the coupler. Consider the case shown in Figure 3.48 where a microprocessor controller is required to drive a relay and a motor. Both the motor and relay can put large voltage spikes on the line, which could cause the microprocessor to act erratically. The optical coupler (OC) will not let this noise from the outside world go back upstream into the controller because the LED is designed to generate light, not receive it.

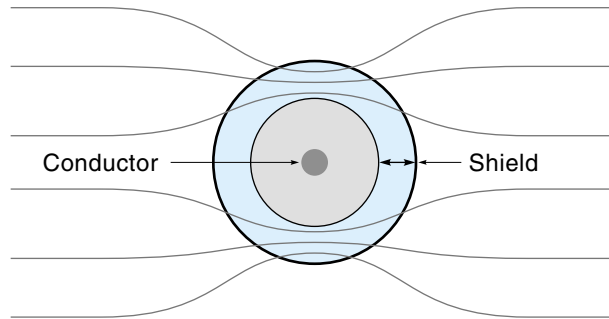
Another use of the optical coupler is high-voltage protection. Because there is no electrical connection between input and output, high voltage cannot pass either way.

**Figure 3.48**  
Interfacing  
components  
with optical  
isolators reduce  
noise spikes.



**Figure 3.49**

Shielding a conductor from a magnetic field. (The shield is shown thicker than normal for illustrative purposes.)



This protects the relatively delicate digital controller circuitry from high voltage spikes in the actuator power lines.

### Shielding

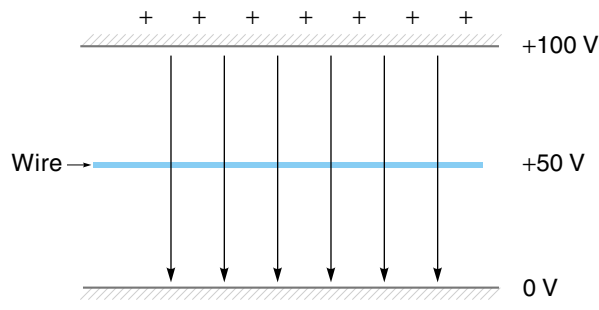
A current through a wire generates a magnetic field around that wire. If another wire happens to be in the vicinity, the magnetic field from the first wire will induce a current in the second wire, proportional to the rate of change of the magnetic field. Therefore, DC will not induce current in a nearby wire, but AC will. This undesired induced current is electrical noise. Good design practice dictates that wires with large AC currents, such as 60-Hz power wires, be kept apart from low-level signal wires.

Shielding cannot block the **magnetic field noise**, but it can draw it away from the signal wire. This is shown in Figure 3.49 in a cross-section view. A shielding material is used that has a low resistance to the magnetic field. For this application, steel is better than copper or aluminum. The magnetic lines of force are more inclined to go through the shield than the air, so the field within the shield is sharply reduced. For this to be effective, the shield must be relatively thick-walled, and in fact, even a heavy conductor placed near the signal wire will act as a magnetic shield.

Another source of noise is from an **electric field**. An isolated wire placed in an electric field will assume a voltage proportional to that field. Figure 3.50 shows a wire

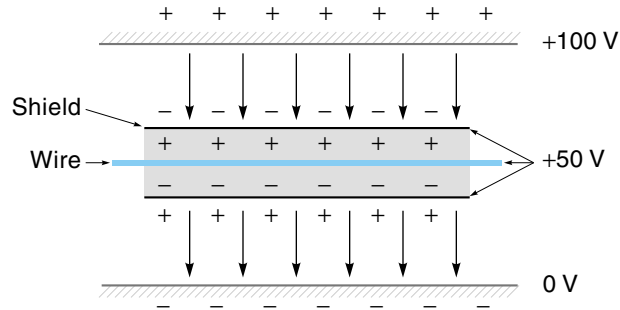
**Figure 3.50**

A wire in an electric field.



**Figure 3.51**

A wire in an ungrounded shield.



in the middle of a 100-V field. The wire will assume a voltage of 50 V. If the field voltage is AC, then an AC voltage will be induced on the wire. Large time-varying electric fields are common in an industrial environment.

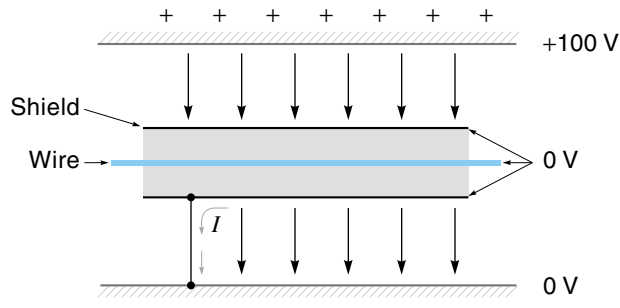
Merely putting a conductive shield around a wire will not keep the field out. As Figure 3.51 shows, the shield itself would acquire a voltage from the field and then reradiate this voltage to the wire. However, if the shield is grounded (forced to be 0 V), then there can be no electric field within the shield, and the wire is protected (Figure 3.52). For best results, the shield should not be used as the signal return path because the shield will probably have noise on it (from the shielding action). The ideal setup is shown in Figure 3.53, where both the signal and the return are contained within a shield and the shield is grounded at one end.

### Shield-Grounding Considerations

The time to think about grounding is when a system is being designed. A grounding strategy built in from the beginning is better than trying to patch things after the system is built. We have discussed the fact that shields must be grounded to be effective and that they should be grounded at only one end. Because the controller is the hub of the control system, it makes sense to tie all shields together at the controller and ground them there. This is called a **single-point ground** and is illustrated in Figure 3.54. It is important that

**Figure 3.52**

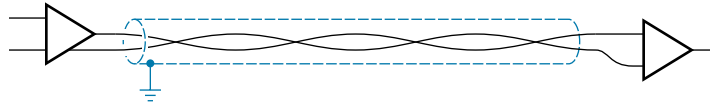
An electrostatic shield grounded at one end.





**Figure 3.53**

A signal and return contained within a shield.



the single-point ground have a solid, low-impedance connection to a real earth ground; otherwise, noise coming in on one shield will reradiate out on all the other shields.

Sometimes the shield cannot be isolated from ground at the component end. In such cases, the best approach is to insert an isolation circuit and break the shield at that point (Figure 3.55). This approach has the advantage of allowing individual components to be grounded for safety reasons, yet preventing ground loops.

## Practical Wiring Considerations

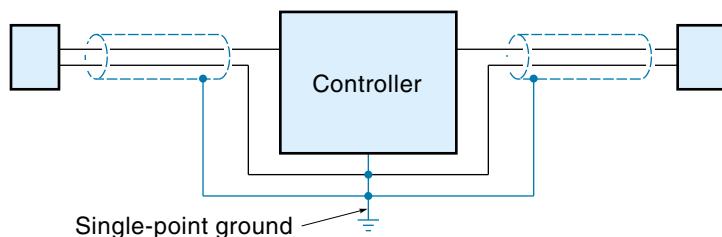
### Wire Size

Most electrical wires are made of copper because it is one of the best conductors available. But even copper has some resistance. This resistance can usually be ignored for small currents and short runs, but for longer runs and bigger currents, wire size becomes an issue. The thicker the wire, the less resistance it has, so a thicker wire can carry more current (than a thinner wire) without heating up, and there will be less voltage drop along its length. Wire size is specified by the American Wire Gage (AWG) shown in Table 3.2. Notice that as the AWG # gets bigger, the diameter gets smaller, the resistance (in ohms per 1000 ft ) goes up, and the current capacity goes down.

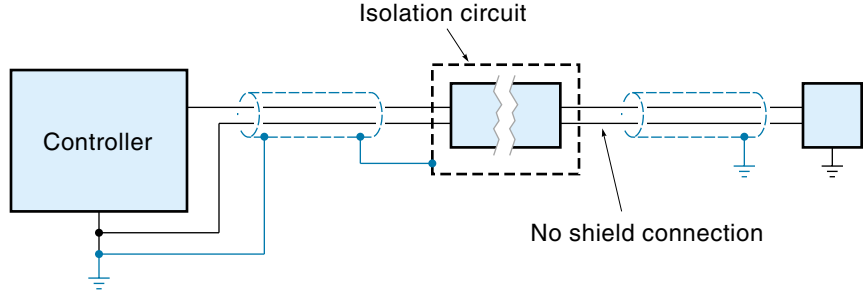
Generally speaking there are two categories of wire usage: power and signal. Power wires supply electrical power (for example, to drive larger motors), and signal wires are used for communication (for example, to connect a limit switch or temperature sensor). Power wires tend to be in the range of AWG #0000 to #14, depending on the current capacity required. Signal wires tend to be in the range of AWG #18 to #28, depending on the voltage drop that can be tolerated. As explained earlier in the chapter, sensors that use the current loop technique are immune to voltage drop problems, so wire size is usually not an issue. However sensors that output a voltage signal have to be concerned with voltage drops in the line. Example 3.11 demonstrates how to determine the correct signal wire size.

**Figure 3.54**

All shields are grounded at one place.



**Figure 3.55**  
Using an isolation circuit to avoid ground loops.



**TABLE 3.2**

**American Wire Gage (AWG) sizes**

AWG #	Diameter (in.) *	$\Omega/1000$ ft at 20°C	Safe current capacity (A)
0000 (4/0)	.528	.0490	230
000 (3/0)	.470	.0618	200
00 (2/0)	.418	.0708	175
0	.372	.0983	150
1	.332	.1240	130
2	.292	.1563	115
3	.260	.1970	100
4	.232	.2485	85
6	.184	.3591	65
8	.1285	.6282	45
10	.1019	.9989	30
12	.0808	1.588	20
14	.0641	2.525	15
16	.0508	4.016	10
18	.0403	6.385	6.5
20	.0320	10.15	4
22	.0253	16.14	2.5
24	.0201	25.67	1.5
26	.0159	40.81	1
28	.0126	64.90	.5
30	.0100	103.2	

\*Diameters for AWG #0000 through #6 are for stranded wire

**EXAMPLE 3.11**

A temperature sensor connects to a controller 100 ft away using AWG#24 wire (see figure 3.56). The sensor puts out a voltage from 0V to 10V. The controller has an input resistance of 1K $\Omega$ . What will be the maximum voltage drop along the wire.

**SOLUTION**

First calculate the resistance of the wire. From the wire table (Table 3.2) we see that #24 wire has a resistance of 25.67 $\Omega$ /1000ft. The total length of wire is 100 ft (100ft out and 100 ft back).

$$\text{Total resistance of wire} = \frac{25.67 \Omega}{1000 \text{ ft}} \times 200 \text{ ft} = 5.13 \Omega$$

From ohms law we know that voltage drop is proportional to current, so we need to determine the maximum expected current in the wire. That will occur when the sensor is putting out 10 V. Note the circuit resistance is the 1K $\Omega$  load plus the wire resistance.

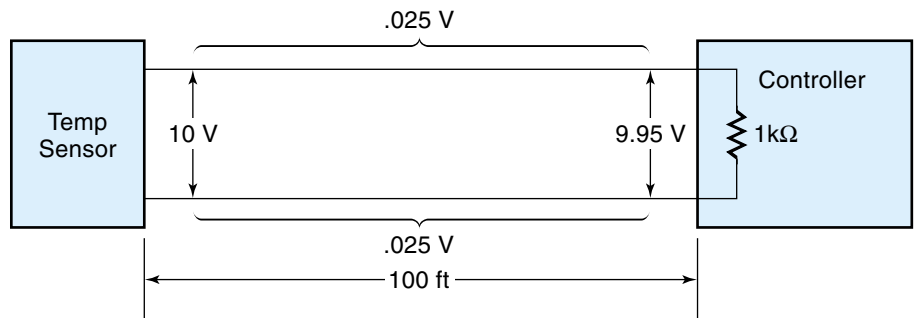
$$\text{Maximum current in wire} = \frac{V}{R} = \frac{10\text{V}}{1000\Omega + 5.13 \Omega} = 9.95 \text{ mA}$$

Knowing the current we can calculate the voltage drop along the wire (using ohms law)

$$\text{Voltage drop along wire} = I \times R = 9.95 \text{ mA} \times 5.13\Omega = .05\text{V}$$

So, in this case the total wire voltage drop is only .05V (.025V out and .025V back) leaving a signal voltage at the controller of 9.95 V (see figure 3.56).

**Figure 3.56**  
Voltage drop along a sensor cable (Example 3.11).

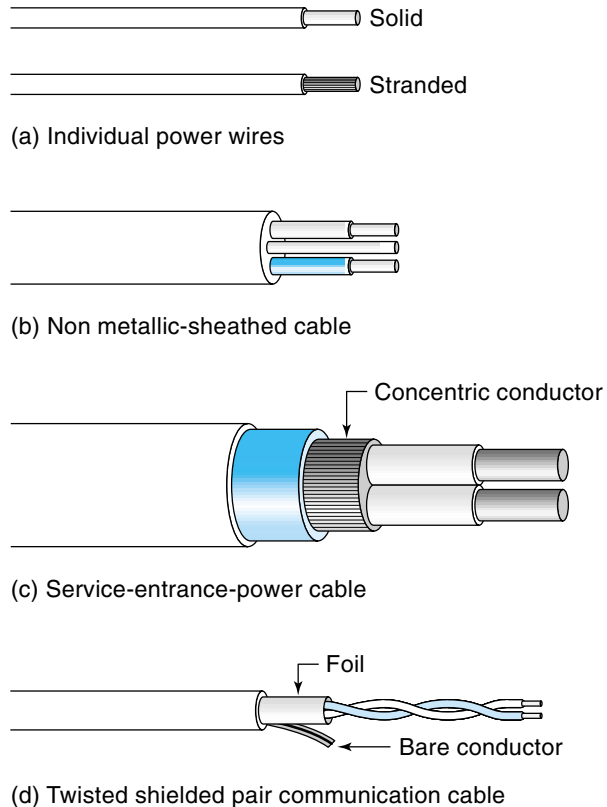


### Wire Types

Power wire is available as individual insulated wires [see Figure 3.57(a)], or in pre-made cables. Probably the most common power cable is the nonmetallic-sheathed copper cable shown in Figure 3.57(b), which is available in sizes #2 through #14. You probably recognize this as standard house wiring cable. Larger power wire is available as service-entrance cable [see Figure 3.57(c)]. This cable is available in sizes 4/0 (0000) through #8, usually as stranded copper or aluminum.

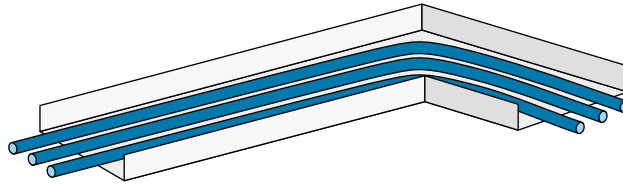
Signal wire is used to connect sensors, switches, and small motors and is typically size #18 to #24. This wire usually comes in pairs that are twisted together, and it may or may not be shielded. Twisting the wires not only keeps them together, but also reduces interference problems. Shielding further reduces susceptibility to interference. A typical twisted shielded pair is shown in Figure 3.57(d).

**Figure 3.57**  
Wire types.

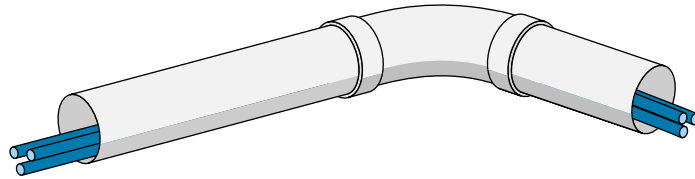


**Figure 3.58**

Cable routing hardware.



(a) Cable tray



(b) Conduit

### *Cable Routing*

The sheer number of cables in an automated industrial setting requires some organization. The individual wires that go out to a sensor or limit switch may just be tied or clipped to whatever superstructure is handy, but at some point in the system, multiple cables going in the same direction will be grouped in a cable tray or electrical conduit (see Figure 3.58). The first rule is to put power and signal lines in separate trays or conduits if at all possible, and if they must cross, they should cross at right angles. Even if the signal wires are in a continuous metallic wire way or conduit, they should be kept at least 3 inches from ac power lines with up to 20 A and at least 6 inches from ac power lines with more than 20 A. If the signal wires are not in a metal conduit or tray, the separation should be doubled.

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## SUMMARY

The operational amplifier (op-amp) is a linear amplifier packaged as an integrated circuit, which has the following desirable properties: (1) very high open-loop gain (typically, 100,000 or more), (2) very high input resistance ( $1\text{M}\Omega$  or more), and (3) low output resistance ( $50\text{--}75\ \Omega$ ). With the addition of a few passive components, many different useful interface circuits can be built from op-amps, such as the following:

- A voltage follower has a voltage gain of 1 but can boost the current and isolate different stages in a circuit.

- Inverting and noninverting amplifiers are stable linear amplifiers where the gain can be set with two resistors.
- A summing amplifier gives an output voltage that is the sum of multiple input voltages.
- An instrumentation amplifier is a practical differential amplifier usually packaged as an IC. It amplifies the difference between two voltages and includes high-resistance buffered inputs and an easily selectable gain.
- Integrators and differentiators give an output voltage that is proportional to the change of input over time.
- An active filter is a circuit using an amplifier that passes certain frequencies and rejects others.
- A comparator compares two analog voltages and identifies which one is larger (with a digital output).

A current loop is a special type of signal interface that uses current instead of voltage to convey the information. This system is immune to wire resistance in the connecting cables because the current is the same everywhere in a closed loop, regardless of the resistance. The current-loop system requires special transmitter and receiver circuits, both of which can be made with op-amps.

The analog switch is a solid-state device that can switch analog signals. A common use of analog switches is for analog multiplexing. An example of multiplexing is where a number of analog sensor signals are connected, one at a time, to an ADC. Another interface device is a sample-and-hold circuit, which can temporarily hold an analog voltage on a charged capacitor until it can be processed.

Earth ground refers to the voltage on the physical earth and is *not* the same as signal return or signal common in a circuit. Equipment is grounded to the earth primarily for safety reasons. However, if the same conductor is grounded at two locations, different local ground voltages can cause large ground-loop currents in the conductor, which is undesirable. To avoid this situation, conductors should be grounded at only one end; if that is not possible, isolation circuits such as optical isolators can be used to break the electrical continuity.

Shielding protects a signal from electrical noise. Electrical noise comes in two forms, magnetic fields and electrical fields, and shielding considerations are different for each type.

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## GLOSSARY

**active filter** A circuit that incorporates an op-amp.

**analog switch** A solid-state device that performs the same function as a low-power mechanical switch.

**band-pass filter** A circuit that allows only a specified range of frequencies to pass and attenuates all others above and below the pass band.

**chatter** the condition that occurs when the output of a comparator oscillates when the input voltage is near the threshold voltage.

**closed-loop gain** The gain of an amplifier when feedback is being used. The value of closed-loop gain is less than open-loop gain (for negative feedback).

**common mode rejection** System whereby a common voltage in both wires of a differential signal is canceled out. This system is used to advantage to reduce noise that is usually common to both wires.

**comparator** A type of op-amp that is used open-loop to determine if one voltage is higher or lower than another voltage.

**current loop** In a signal transmission system, a single loop of wire that goes from the transmitter to the receiver and back to the transmitter. The signal intelligence is conveyed by the current level instead of the voltage. This system is immune to voltage drops caused by wire resistance.

**current-to-voltage converter** An op-amp based circuit used as a receiver for a current-loop system.

**cutoff frequency** The frequency at which a filter circuit's gain drops to 0.707 (or -3 db. )

**DC offset voltage** The small voltage that may occur on the output of an op-amp, even when the inputs are equal; DC offset can be eliminated with a resistor adjustment.

**decibel** Gain expressed in a logarithmic scale.

**differential amplifier** A circuit that produces an output voltage that is proportional to the instantaneous voltage difference between two input signals. Op-amps by themselves are difference amplifiers; however, a practical differential amplifier incorporates additional components.

**differential voltage** A signal voltage carried on two wires, where neither wire is at ground potential.

**differentiator** An op-amp circuit that has an output voltage proportional to the instantaneous rate of change of the input voltage.

**earth ground** The voltage at (or connection to) the surface of the earth at some particular place.

**electric field** A condition that exists in the space between two objects that are at a different voltage potential. A wire in this space will assume a "noise" voltage proportional to the field strength.

**high-pass filter** A circuit that allows higher frequencies to pass but attenuates lower-frequency signals.

**instrumentation amplifier** A practical differential amplifier, usually packaged in an IC, with features such as high input resistance, low output resistance, and selectable gain.

**integrator** An op-amp based circuit that has an output voltage proportional to the area under the curve traced out by the input voltage.

**interfacing** The interconnection between system components.

**inverting amplifier** A simple op-amp voltage amplifier circuit with one input, where the output is out of phase with the input; one of the most common op-amp circuits.

**inverting input** The minus (–) input of an op-amp; the output will be out of phase with this input.

**isolation circuit** A circuit that can transfer a signal voltage without a physical electrical connection.

**low-pass filter** A circuit that allows lower-frequency signals to pass but attenuates higher-frequency signals.

**magnetic field noise** An unwanted current induced in a wire because the wire is in a time-varying magnetic field.

**multiplexing** The concept of switching input signals (one at a time) through to an output; is typically used so that multiple sensors can use a single ADC (analog-to-digital converter).

**negative feedback** A circuit design where a portion of the output signal is fed back and subtracted from the input signal. This results in a lower but predictable gain and other desirable properties.

**noninverting amplifier** A simple op-amp voltage amplifier circuit with one input, where the output is in phase with the input.

**noninverting input** The positive (+) input of an op-amp; the output will be in phase with this output.

**notch filter** A circuit that attenuates a very narrow range of frequencies.

**operational amplifier (op-amp)** A high-gain linear amplifier packaged in an integrated circuit; the basis of many special-purpose amplifier designs.

**open-loop gain** The gain of an amplifier when no feedback is being used; it is usually the maximum gain possible.

**optical coupler** An isolation circuit that uses a light-emitting diode (LED) and a photocell to transfer the signal.



**programmable gain instrumentation amplifier** An instrumentation amplifier with fixed gains that can be selected with digital inputs.

**sample-and-hold circuit** A circuit that can temporarily store or remember an analog voltage level.

**signal common** The common voltage reference point in a circuit, usually the negative terminal of the power supply (or battery).

**signal return** *See* **signal common**.

**single-ended voltage** A signal voltage that is referenced to the ground.

**single-point ground** A single connection point in a system, where all signal commons are connected together and then connected to an earth ground.

**summing amplifier** An op-amp circuit that has multiple inputs and one output. The value of the output voltage is the sum of the individual input voltages.

**virtual ground** A point in a circuit that will always be practically at ground voltage but that is not physically connected to the ground.

**voltage follower** A very simple but useful op-amp circuit with a voltage gain of 1, used to isolate circuit stages and for current gain.

**window comparator** A comparator with built-in hysteresis, that is, two thresholds: an upper switch-in point and a lower switch-in point.

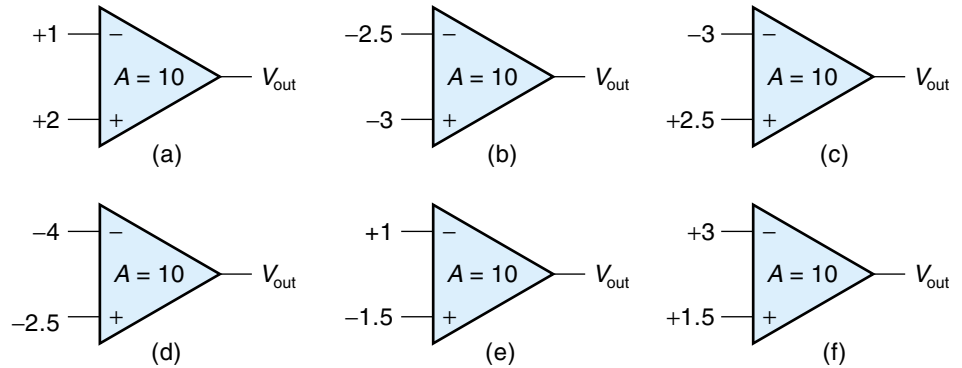
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## EXERCISES

### Section 3.1

1. Find  $V_{\text{out}}$  for each amplifier in Figure 3.59. Assume the gain is 10 in all cases.
2. An op-amp has an open-loop gain of 200,000 and is being supplied with +12 V and -12 V. What is the maximum voltage difference that can exist between  $V_1$  and  $V_2$  before saturation? (Is this why  $V_1$  and  $V_2$  are considered “virtually” the same?)
3. Use the basic op-amp assumptions to describe how a voltage follower works.
4. Draw the circuit of an inverting amplifier with a gain of 35.
5. Draw the circuit of an inverting amplifier with a gain of 50 and show the pin numbers of the 8 pin 741 general-purpose op-amp. Power supply voltage is  $\pm 15$  V.
6. Draw the circuit of a noninverting amplifier with a gain of 24.
7. Draw the circuit of a noninverting amplifier with a gain of 30 and draw the pin numbers of the 8 pin 741 general-purpose op-amp. Power supply voltage is  $\pm 15$  V.
8. Draw the circuit of a summing amplifier with four inputs. The gain for each input should be 15. Assume the impedance of each source connected to the summing amp is 1 k $\Omega$ .

Figure 3.59



9. Draw the circuit of a difference amplifier with a gain of 20. Assume the impedance of each source is  $6 \text{ k}\Omega$ .
10. Sketch the output of an integrator circuit that has an input waveform shown in Figure 3.60 (assume  $RC = 1$ ):
11. Sketch the output of an integrator circuit that has  $R = 2 \text{ M}\Omega$  and  $C = 10 \text{ }\mu\text{F}$ . The input waveform is shown in Figure 3.60.
12. Sketch the output of a differentiator circuit that has an input waveform as shown in Figure 3.61 (assume  $RC = 1$ ).
13. Sketch the output of a differentiator circuit that has  $R = 2 \text{ M}\Omega$  and  $C = 10 \text{ }\mu\text{F}$ . The input waveform is shown in Figure 3.61.
14.
  - a. Find the gain in decibels of an amplifier that has a straight voltage gain of 7000.
  - b. Estimate the gain of the same amplifier using the method of Example 3.8.
15.
  - a. Find the gain in decibels of an amplifier has a straight voltage gain of 3300.
  - b. Estimate the gain of the same amplifier using the method of Example 3.8.
16.
  - a. Find the straight voltage gain of an amplifier that has a gain of 35 db.
  - b. Estimate the straight gain of the same amplifier using the method of Example 3.9.
17.
  - a. Find the straight voltage gain of an amplifier that has a gain of 50 db.
  - b. Estimate the straight gain of the same amplifier using the method of Example 3.9.

Figure 3.60

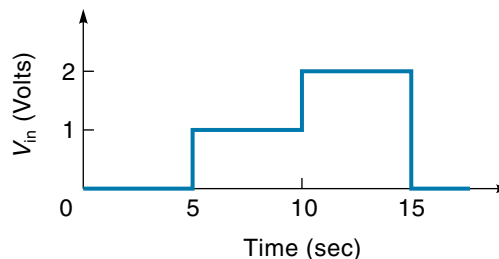
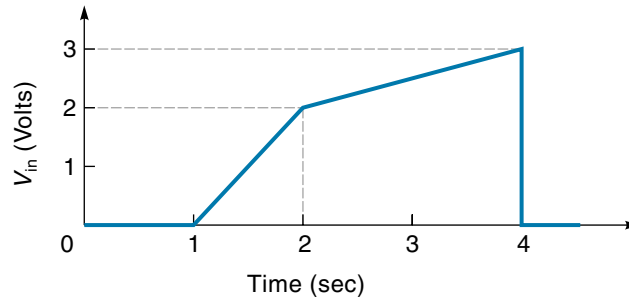


Figure 3.61



18. Draw the circuit of a low-pass filter with  $f_c = 1$  kHz and a gain of 10. Use  $R = 1$  k $\Omega$ .
19. Draw the circuit of a low-pass filter with  $f_c = 5$  kHz and a gain of 20. Use  $R = 1$  k $\Omega$ .
20. Draw the circuit of a high-pass filter with  $f_c = 10$  kHz and a gain of 10. Use  $R = 1$  k $\Omega$ .
21. Draw the circuit of a high-pass filter with  $f_c = 8$  kHz and a gain of 15. Use  $R = 1$  k $\Omega$ .
22. Draw the circuit of a notch filter with  $f_N = 1.6$  kHz. Use  $R = 2$  k $\Omega$ .
23. Draw the circuit of a comparator that will switch from 0 to 5 V when the analog input goes above 2.9 V.

### Section 3.2

24. Explain why the signal is never attenuated in a current-loop system.
25. Draw the circuit of a current-loop system. When  $V_{in}$  is 1 V, the loop current should be 5 mA. At all times,  $V_{out}$  should equal  $V_{in}$ .
26. What is an *analog switch*, and how is it different than a mechanical switch?
27. The sample-and-hold circuit of Figure 3.42 is being used to hold a signal for 0.25. If  $R_s = 1$  k $\Omega$ , what size capacitor is needed?

### Section 3.3

28. Explain the difference between *earth ground* and *signal return*.
29. What is a *ground loop*? How is it possible to have large currents in the return line when there is no apparent power supply?
30. Find the current in a ground loop that is 200 ft long, has a wire resistance of 0.05  $\Omega$ /ft, and has a difference of 4 V between grounds.
31. Find the current in a ground loop that is 300 ft long, has a wire resistance of 0.01  $\Omega$ /ft, and has a difference of 5 V between grounds.
32. Why is it important to have the metal chassis of equipment connected to earth ground?
33. Describe how the optical isolator in Figure 3.47 works.

34. What are the reasons for using isolation circuits in the interconnection of a control system?
35. Explain the principle behind magnetic shielding.
36. Why is it necessary to ground an electrostatic shield?
37. Explain the concept of single-point ground and why it is important.
38. Find the total voltage drop in a 200-ft #20 twisted pair wire circuit carrying 100 mA. (Hint: The signal has to go out 200 ft and back.)
39. Find the total voltage drop in a 500-ft #22 twisted pair wire circuit carrying 50 mA. (Hint: The signal has to go out 500 ft and back.)

# CHAPTER 4

## Switches, Relays, and Power-Control Semiconductors

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### OBJECTIVES

After studying this chapter, you should be able to:

- Understand mechanical switch types, configurations, and terms.
- Describe the types and operation of electromechanical relays.
- Understand the characteristics of solid-state relays and the advantages and disadvantages of solid-state relays compared with electromechanical relays.
- Comprehend the operation of power transistors and power amplifiers, calculate transistor circuit currents, and select a replacement power transistor.
- Understand basic JFET and power MOSFET characteristics and applications.
- Understand the characteristics and operation of silicon-controlled rectifiers and basic silicon-controlled rectifier circuits.
- Understand how a triac operates and some triac applications.
- Explain the use of some trigger circuits for the silicon-controlled rectifiers and triac, in particular those that use the unijunction transistor, programmable unijunction transistor, or diac.

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### INTRODUCTION

An important ability of almost all control systems is to regulate the flow of electrical power because electricity is usually the link between the controller and the actuator (which is typically an electric motor). The problem is that the small-signal output of the controller is usually not large enough to drive the load directly and must be amplified in some way.

Three classifications of components are used to control electrical power: (1) electro-mechanical devices such as switches and relays, (2) power transistors and field effect

transistors, and (3) a group of semiconductor devices called thyristors, which includes silicon-controlled rectifiers and triacs. All these power control devices are the subject of this chapter.

## 4.1 SWITCHES

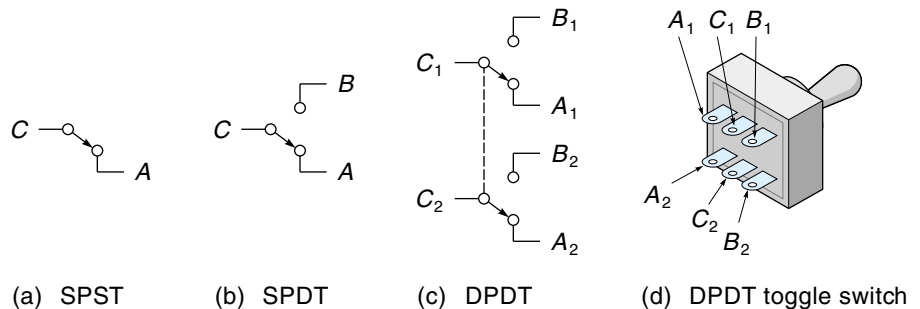
A *mechanical switch* is a device that can open or close, thereby allowing a current to flow or not. As you have no doubt observed, switches come in many different shapes, sizes, and configurations.

### Toggle Switches

Probably the most common switch type is the **toggle switch**, which is available in various contact configurations. Each switch consists of one or more *poles*, where each pole is actually a separate switch. The contact arrangement for the **single-pole/single-throw switch** (SPST), the simplest switch, is illustrated in Figure 4.1(a). Notice that it has a single set of contacts that can either open or close. Next, up the line of complication, is the **single-pole/double-throw switch** (SPDT) illustrated in Figure 4.1(b). Notice that the movable “arm” called the *common* (C), or **wiper**, can connect with either contact A or B. Figure 4.1(c) illustrates a **double-pole/double-throw switch** (DPDT), which consists of two electrically separate SPDT switches in one housing that operate together. In Figure 4.1(d), notice how the terminals are arranged on the back of the DPDT switch housing, with the three terminals for each pole running the “long way” on the switch body. Although not as common, switches with up to six poles are available.

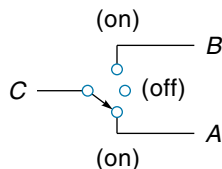
So far the discussion has centered on the **two-position switch**, which comes in two configurations: the simple *on-off* [Figure 4.1(a)] and the *on-on* [Figure 4.1(b)]. A **three-position switch**, called *on-off-on*, has a stable position in the middle, where the C contact is not connected to either A or B, as illustrated in Figure 4.2. Such a switch could be used to control a motor, where the three positions might correspond to “forward,” “off,” and “reverse.”

**Figure 4.1**  
Toggle switch contact arrangements.



**Figure 4.2**

A three-position switch: on–off–on.



Another variation of the toggle switch is to have one or more of the positions be spring-actuated, meaning that physical pressure must be maintained on the handle to keep it in position. These are called **momentary-contact switches**, an example of which is a car's ignition switch where you must maintain continuous pressure on the key to engage the starter motor. Table 4.1 shows some of the variations possible for a particular style of toggle switch. For example, the third switch (1SFX191), is a three-position switch with one of the on positions being momentary.

**TABLE 4.1****Typical Toggle Switches\***

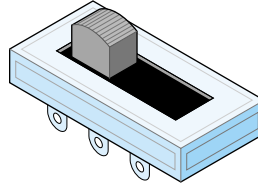
Type	Number of poles	Circuit
1SBX191	1	On–Off–On
1SCY191	1	On On
1SFX191	1	On–Off–On <sup>†</sup>
1SGX191	1	On On <sup>†</sup>
1SHX191	1	On <sup>†</sup> –Off–On <sup>†</sup>
2SBX191	2	On–Off–On
2SCY191	2	On On
2SFX191	2	On–Off–On <sup>†</sup>
2SGX191	2	On On <sup>†</sup>
2SHX191	2	On <sup>†</sup> –Off–On <sup>†</sup>
3SBX191	3	On–Off–On
3SCY191	3	On On
3SFX191	3	On–Off–On <sup>†</sup>
3SCX191	3	On On <sup>†</sup>
3SHX191	3	On <sup>†</sup> –Off–On <sup>†</sup>
4SGX191	4	On–Off–On
4SCY191	4	On On
4SFX191	4	On–Off–On <sup>†</sup>
4SGX191	4	On On <sup>†</sup>
4SHX191	4	On <sup>†</sup> –Off–On <sup>†</sup>

\*Rated 5 A at 125 Vac; 5 A at 28 Vdc.

<sup>†</sup>Momentary contact.

**Figure 4.3**

A slide switch.



Toggle switch contacts have maximum voltage and current ratings for AC and DC operation. A typical toggle switch can handle less DC voltage than AC voltage at the rated current; for example, a switch might be rated at 5 A for 125 Vac or 5 A for 28 Vdc. The reason for the difference is that when the contacts are just opening, the current will continue (briefly) to arc across, which tends to burn and pit the contact surface. AC will arc less than DC for the same voltage because AC is going to 0 V twice each cycle. Also, most toggle switches have a “snap” action which pulls the contacts apart rapidly to minimize arcing.

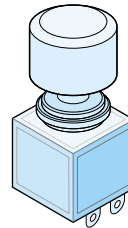
Closely related to the toggle switch is the **slide switch** illustrated in Figure 4.3. Although mechanically different internally, the slide switch performs the same functions as the toggle switch and is available in the same configurations. Slide switches tend to be less expensive but are not available with the high current rating of toggle switches.

### Push-Button Switches

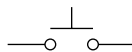
**Push-button switches** [Figure 4.4(a)] are almost always the momentary type—pressure must be maintained to keep the switch activated. Figure 4.4(b-d) shows the symbol for the push-button switch. Notice there are two configurations possible: **normally open**

**Figure 4.4**

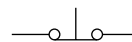
Push-button switches.



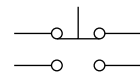
(a) Push-button switch



(b) Normally open (NO) switch



(c) Normally closed (NC) switch

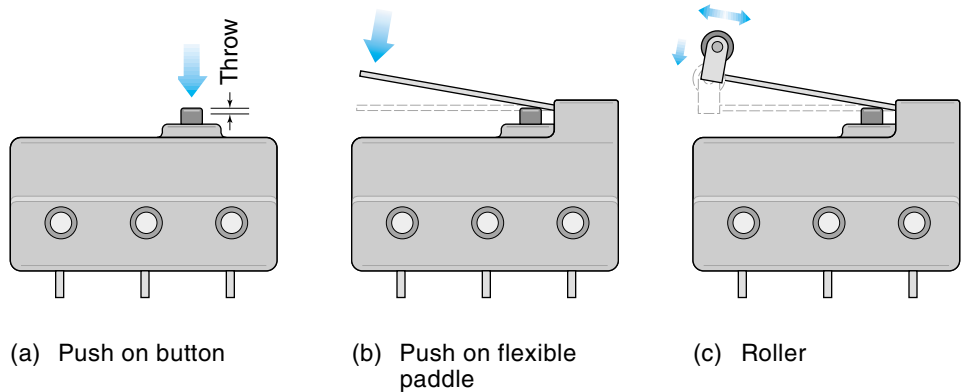


(d) NC and NO switch



**Figure 4.5**

Limit switches (using a microswitch type).



(NO) and **normally closed** (NC). For the NO switch [Figure 4.4(b)], the contacts are open until the button is pushed; and for the NC switch [Figure 4.4(c)], the contacts are closed when the switch is “at rest” and open when the button is pushed.

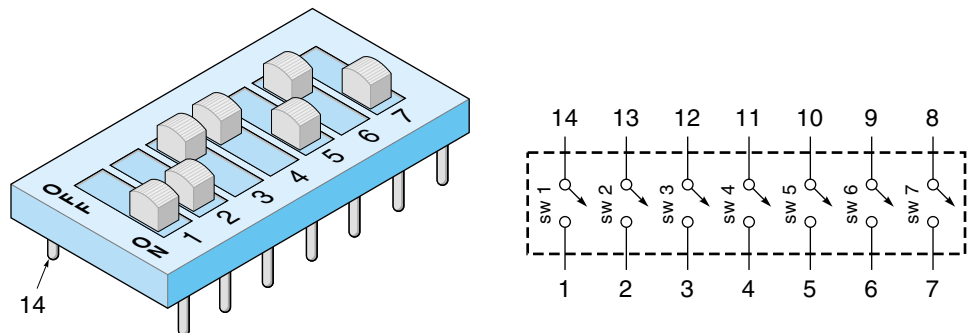
### Other Switch Types

A **limit switch** is a push-button switch mounted in such a position that it is activated by physical contact with some movable object. An example is a car door switch, which senses whether or not the door is closed. Limit switches are available with different kinds of actuator hardware, such as the “paddle” or roller. Often these are mounted on a small standard-sized switch body called a **microswitch** (developed by the MicroSwitch Company). A microswitch requires a very small throw of a few thousandths of an inch. Figure 4.5 illustrates some examples of limit switches. (Limit switches and other proximity sensors are discussed in Chapter 6.)

A **DIP switch** is a set of small SPST switches built into a unit shaped like an integrated circuit (IC) (DIP stands for dual in-line package). The DIP switch can be plugged into an IC socket or soldered into a circuit board. As shown in Figure 4.6,

**Figure 4.6**

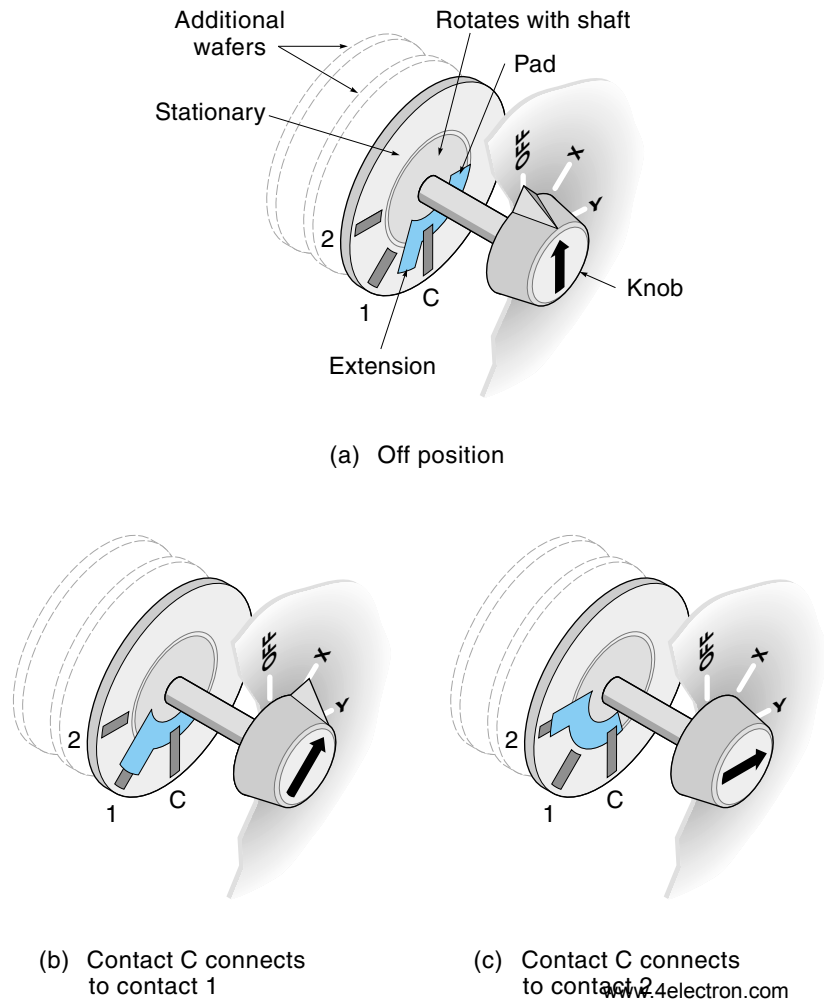
A DIP switch.



each individual switch uses two pins directly across from each other; that is, switch 1 uses pins 1 and 14, switch 2 uses pins 2 and 13, and so on. (Do not use a pencil to set the switches; the graphite can work its way into the switch and short it out.)

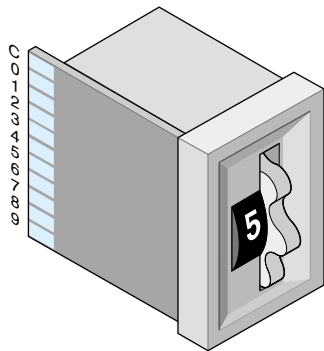
A **rotary switch**, considerably different from the switches discussed so far, can perform multiple, complicated switching operations. As shown in Figure 4.7, the rotary switch is constructed of **switch wafers** mounted along a single shaft. The inner part of each wafer rotates in steps with the shaft, and the outer part remains stationary. To understand how the switch works, consider Figure 4.7(a), which shows the switch in the off position. In this position, the C terminal is making contact with the pad, but the pad is not touching either terminals 1 or 2. In Figure 4.7(b), the shaft has been rotated clockwise to position X. Notice that the C terminal is still connected to the pad, but now the little extension of the pad is making contact with terminal 1. In Figure 4.7(c), the shaft

**Figure 4.7**  
Rotary switch  
operation.

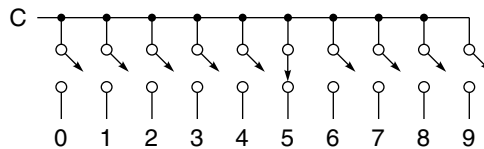


**Figure 4.8**

A thumbwheel switch (ten-decimal position).



(a) Switch body



(b) Thumbwheel switch schematic

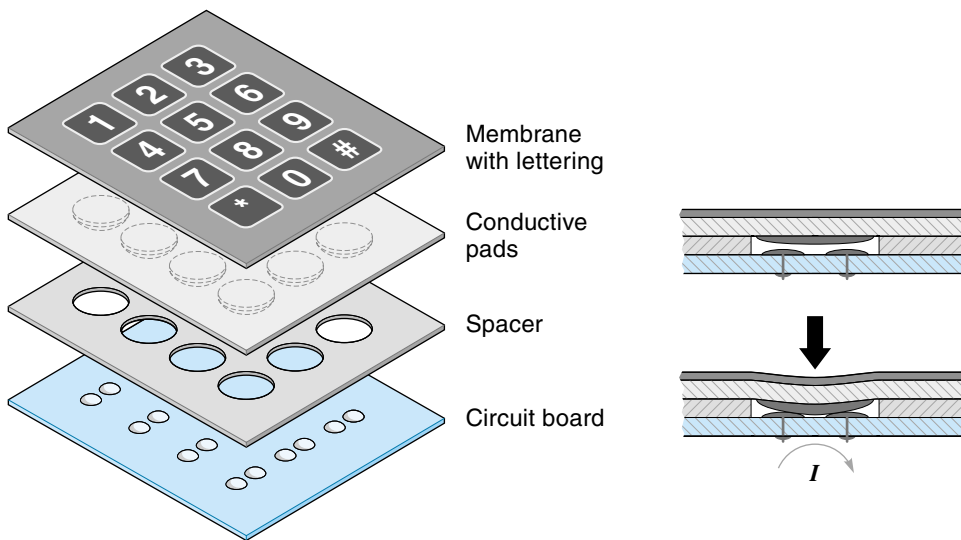
has been rotated to position *Y*, and now the pad is in contact with terminal 2. Notice that there could be three or four such switches on the same wafer, as well as multiple wafers.

A **thumbwheel switch**, a special type of rotary switch, is used to input numeric data. The operator selects a number by rotating the numbered wheel (Figure 4.8), and each number corresponds to a different switch setting. The switch schematic in Figure 4.8(b) shows that one of ten separate terminals is connected with the *C* terminal. Also available are thumbwheel switches that output 4-bit BCD (binary coded decimal) and other codes.

Another type of data-input device is the **membrane switch** keypad. This type of keypad consists of push-button switches built up from a number of layers, as shown in Figure 4.9(a). The bottom layer is usually a printed-circuit board with two nonconnecting pads

**Figure 4.9**

A membrane switch.



(a) Construction

(b) Switch operation

for each switch. Lying over the circuit board is a spacer layer with a hole over each switch position. The third layer is flexible and has a conductive pad over each switch. Overlying the whole thing is a flexible waterproof membrane, which has the key lettering. When a key is pressed, the conductive pad is forced into contact with the circuit board and provides a path for the current, as illustrated in Figure 4.9(b). Membrane switches are especially appropriate in dirty industrial environments because the membrane keeps dirt from getting into the switch assembly.

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## 4.2 RELAYS

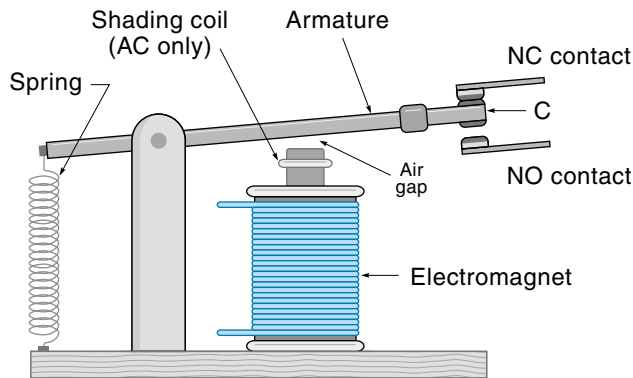
### Electromechanical Relays

The **electromechanical relay** (EMR) is a device that uses an electromagnet to provide the force to close (or open) switch contacts, in other words, an electrically powered switch. A diagram of a simple relay is shown in Figure 4.10(a). When the electromagnet, called the *coil*, is energized, it pulls down on the spring-loaded *armature*. Relay contacts are described as being one of two kinds: normally open contacts (NO), which are open in the unenergized state, and normally closed contacts (NC), which are closed in the unenergized state. Figure 4.10(b) shows one common schematic symbol for the relay. By convention, the symbol always depicts the relay in the unenergized state, so you can easily determine which are the NC and NO contacts from the schematic. Another schematic representation for the relay is shown in Figure 4.10(c). This symbol is used in a type of drawing called a *ladder diagram*. In a ladder diagram, the relay coil and its contacts are separated on the drawing. (Ladder diagrams are discussed in Chapters 9 and 12.)

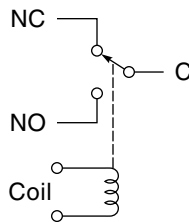
The electrical specifications for the contacts are different than for the coil. For the contacts, the maximum current and voltage for DC and AC operation is specified. For the coil, the intended voltage and coil resistance are usually specified. You can see these numbers on the specification (spec) sheet shown in Table 4.2. The coil voltage and resistance can be used to calculate the steady-state coil current. Actually, it takes more voltage and current to pull in the relay contacts than it does to hold them there because the armature must be pulled in across an air gap. Hence, these quantities are called, respectively, **pull-in voltage** and **pull-in current**. For example, the contacts of a particular 6-V relay actually close at 2.1 V and stay closed until the voltage is decreased to 1 V. The values of voltage and current needed to keep the relay energized are called the **minimum holding voltage** and **sealed current**. Notice that the actual pull-in voltage is much less than the rated coil voltage. This is to guarantee that the relay will pull in quickly and reliably when operated at the rated voltage. Coil voltages are specified to be AC or DC. The difference is that AC coils are constructed with **shaded poles** to prevent “buzzing” with 60-cycle power. A shaded-pole relay has a metal ring around the pole face of the electromagnet (see Figure 4.10). Magnetic flux induced into this ring keeps the relay closed when the AC cycles through 0 V.

**Figure 4.10**

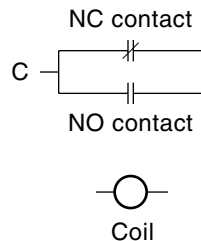
An electromagnetic relay.



(a) Parts of the relay



(b) A common schematic symbol



(c) Schematic symbol for a ladder diagram

**TABLE 4.2**

**Typical General-Purpose Relays\***

Type	Input	Coils		Action
			Ohm	
Y1-SS1.0K	6 DC		1,000	SPDT
Y1-SS220	3 DC		220	SPDT
Y2-V52	6 DC		52	2PDT
Y2-V185	12 DC		185	2PDT
Y2-V700	24 DC		700	2PDT
Y2-Y2.5K	48 DC		2,500	2PDT
Y2.15K	115 DC		15,000	2PDT
Y4.V52	6 DC		52	4PDT
Y4-V185	12 DC		185	4PDT
Y4-V700	24 DC		700	4PDT
Y4-V2.5K	48 DC		2,500	4PDT
Y4-V15K	115 DC		15,000	4PDT
Y6-V25	6 DC		25	6PDT
Y6-V90	12 DC		90	6PDT
Y6-V430	24 DC		430	6PDT
Y6-V1.5K	48 DC		1,500	6PDT
Y5-V9.0K	115 DC		9,000	6PDT

\*Contacts: 2 A typically, 3 A maximum 125 Vac of 28 Vdc.

Relays are available in a variety of sizes, contact configurations, and power-handling capabilities. Some miniature relays can plug into an IC socket and be powered directly from a digital logic gate. On the other end of the spectrum, a power relay, often called a **contactor**, is used to switch the current directly to larger machines and may handle 50 A. Figure 4.11 shows a selection of different relays.

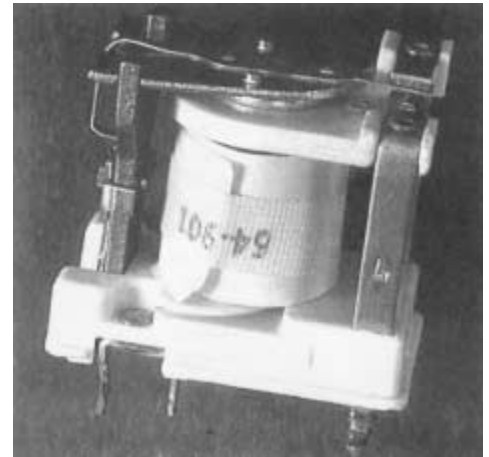
It is well to remember that for two reasons, relays have a finite life. First, because the relay is a mechanical device, the moving parts eventually wear out, and second, the electrical contacts can become pitted because of arcing. The contact wear is very

### Figure 4.11

A selection of relays.  
[Photos (a) and (d)  
are Allen Bradley  
products courtesy  
of Rockwell  
Automation.]



(a) General purpose relay



(b) General purpose relay



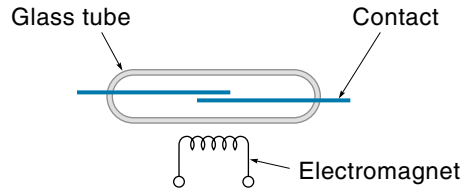
(c) High current relay



(d) Industrial relay

**Figure 4.12**

A reed relay.



dependent on the electric current that is being switched. For example, a certain relay is rated for 9 million operations at 1.5 amps but only 2 million operations at 3 amps. Two million operations sounds like a lot, but if this relay were in a 24-hour factory being used in a repetitive operation every 10 seconds, it would have to be replaced every 8 months. Contact life also depends on the type of load being controlled. For example, inductive loads such as motors cause much more arcing and pitting than resistive loads such as lights and heating elements.

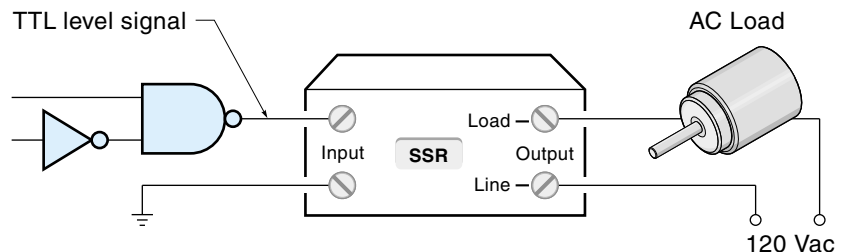
The **reed relay** is unique because the small reedlike contacts are encapsulated in a small sealed glass tube that is evacuated or filled with an inert gas like dry nitrogen. The contacts are activated by an external magnetic field, as shown in Figure 4.12. Contacts are either dry or mercury-wetted. Mercury-wetted contacts have a thin coating of mercury that fills in surface irregularities, making a larger conduction area, and reduces pitting. Generally, reed relays have a long life and low coil voltage (frequently TTL compatible), and are immune to dirty environments; however, they are generally low power (contacts rated at 2 A or less) and vibration sensitive.

### Solid-State Relays

A **solid-state relay** (SSR) is a purely solid-state device that has replaced the EMR in many applications, particularly for turning on and off AC loads such as motors. Physically, the SSR is packaged in a box (about the same size as an EMR), with four electrical terminals, as shown in Figure 4.13(a). The two input terminals are analogous

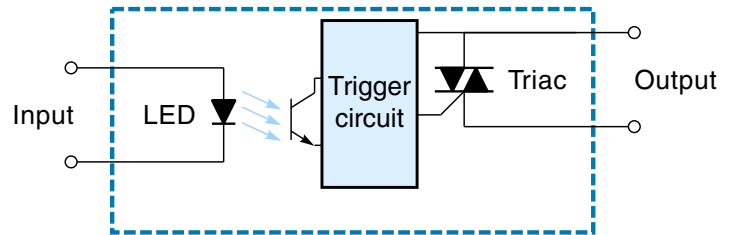
**Figure 4.13**

A solid-state relay.



(a) Using an SSR to drive a motor

**Figure 4.13**  
(cont.)



**(b)** SSR circuit

to the coil of an EMR, and the two output terminals are analogous to the contacts of the EMR (usually SPST, normally open).

The input or *control voltage* of the SSR is typically 5 Vdc, 24 Vdc, or 120 Vac. The 5-Vdc models are designed to be driven directly from TTL digital logic circuits. Turning our attention to the output side of the SSR, we see that the load is placed in series with the 120-Vac or 240-Vac power. The output current can be as high as 50 A in some models. Many SSRs incorporate a feature called **zero-voltage switching**: The line current is switched on at the precise time that the AC voltage is crossing 0 V. This eliminates sharp output voltage-rise times and so minimizes electromagnetic interference noise (EMI).

Figure 4.13(b) shows a block diagram of the interior of an SSR. The input voltage drives an LED, and the light from the LED turns on a photo transistor, which in turn turns on the triac (a solid-state switching device discussed later in this chapter). The LED electrically isolates the input and output sections of the SSR. This is important for two reasons: First, it allows the control electronics to have a separate ground from the power lines; second, it prevents high-voltage spikes in the power circuit from working their way back upstream to the more delicate control electronics.

Solid-state relays have a number of important advantages over electromechanical relays. Having no moving parts means that (theoretically) they will never wear out and makes them practically immune to shock and vibration. Also, because of the built-in electronics, they can be driven with a low-voltage source (such as TTL) regardless of the output-current capability. The main disadvantages of the SSR are the following: (1) They can be “false triggered” by electrical noise; (2) even when on, the output resistance is not exactly 0 ohm, so there is some small voltage drop and consequent power loss within the relay, and when off they may have lethal levels of leakage current; (3) although they are long-lasting, unlike an EMR they do not fail predictably; (4) contact arrangements are limited, so they may not work for all relay applications.

The **hybrid solid-state relay** is similar to the SSR but uses a low-voltage, fast-acting reed relay instead of an LED to turn on the output triac. Using the reed relay provides good electrical isolation and may work better than the SSR in some situations.



## 4.3 POWER TRANSISTORS

**Power transistors** are used extensively in control circuits as both switches and power amplifiers. Power transistors are fundamentally the same as small-signal transistors but are designed to carry more current. Many readers of this text will have already studied transistor operation, but for those who haven't (and perhaps as a review for those who have) the following basic material on transistors is presented. Also, the reader is reminded that this text uses *conventional current flow*, which means that the current is assumed to flow from the positive battery terminal to the negative. If you are familiar with *electron flow*, remember that the theory and “numbers” are the same, only the indicated *direction* of the current is opposite from what you are used to.

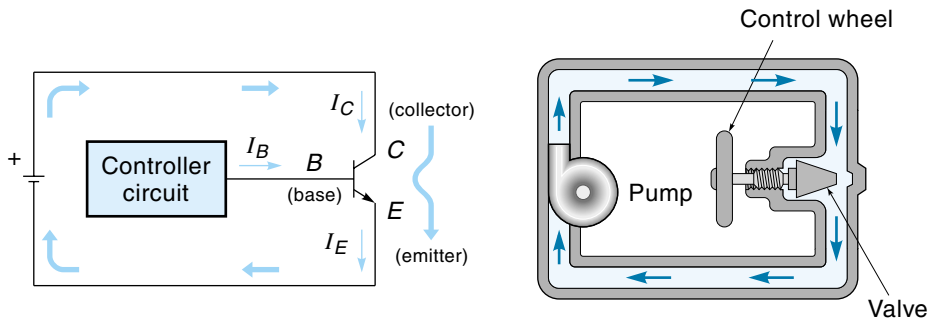
### Bipolar Junction Transistors

The **bipolar junction transistor** (BJT), referred to as a *transistor*, is a three-terminal solid-state device that operates on electric current much like a valve does on water in a pipe. This concept is illustrated in Figure 4.14. Figure 4.14(a) shows that the transistor has three terminals; the **base**, **emitter**, and **collector** and that it is connected into a simple series circuit with the (conventional) current entering the collector (C), and leaving through the emitter (E). Functionally, this is analogous to the system illustrated in Figure 4.14(b) where water is being pumped through a partially open valve. The flow of water can be regulated by opening or closing the valve. In the transistor circuit, the flow of current ( $I_C$ ) can be regulated by adjusting a small control current in the base ( $I_B$ ); the more  $I_B$ , the more  $I_C$ . In fact,  $I_C$  could be 100 times (or more) larger than  $I_B$ .

There are two basic types of transistors: **NPN** and **PNP**, both of which are made from three layers of semiconductor material. As illustrated in Figure 4.15, the only functional difference between the two types is the direction of current flow. The arrowhead of the emitter indicates the (conventional) current direction. The NPN type is more common and will be the type used in the following discussions.

**Figure 4.14**

Basic transistor action (conventional current).

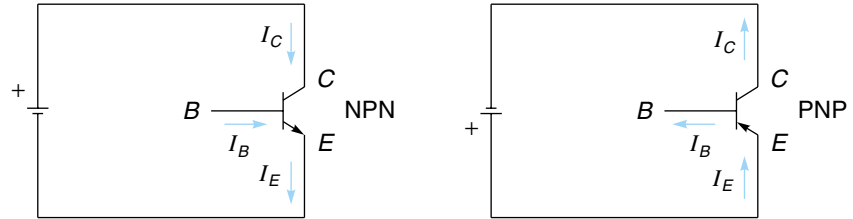


(a) Simple transistor circuit

(b) Water-in-pipe analogy

**Figure 4.15**

NPN and PNP transistors (conventional current).



Basic transistor operation can be summarized with the following statements:

1. Under the proper operating conditions,  $I_C$  will be some multiple of  $I_B$ ; in other words, *the transistor is a current amplifier*. The **forward current gain** ( $h_{FE}$  or  $\beta$ ) (which varies according to transistor type) is expressed as

$$h_{FE} = \frac{I_C}{I_B} \quad (4.1)$$

where

$h_{FE}$  = forward current gain

$I_C$  = collector current

$I_B$  = base current

2. Inside the transistor, the small base current joins with the collector current to become the emitter current (or the emitter current divides to become the base and collector currents). This is expressed as

$$I_E = I_C + I_B \quad (4.2)$$

where

$I_E$  = emitter current

$I_C$  = collector current

$I_B$  = base current

However, because the collector current is usually so much larger than the base current, a useful approximation is to consider the collector current to be equal to the emitter current:

$$I_E \approx I_C \quad (4.3)$$

3. The transistor dissipates power anytime there is a current flowing through it *and* a voltage is across it, expressed as

$$P_D = I_C V_{CE} \quad (4.4)$$

where

$P_D$  = power dissipated by the transistor

$I_C$  = collector current

$V_{CE}$  = voltage between the collector and emitter

**EXAMPLE 4.1**

A power transistor with an  $h_{FE}$  of 50 is operating with a load current ( $I_C$ ) of 3 A. What is the base current?

**SOLUTION**

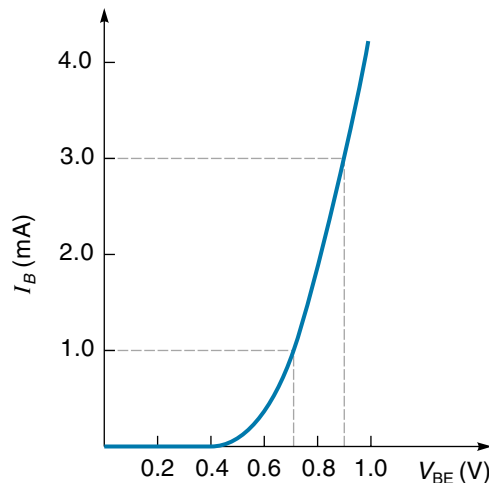
Recall that a transistor is basically a current amplifier where the  $I_C$  is  $h_{FE}$  times larger than the base current ( $I_B$ ). By rearranging Equation 4.1, we can solve for  $I_B$ :

$$I_B = \frac{I_C}{h_{FE}} = \frac{3 \text{ A}}{50} = 60 \text{ mA}$$

We have established that the transistor will “amplify” its base current, but how do you know what the base current is? The base-emitter junction acts like a forward-biased diode, meaning it must first be given a **forward-bias voltage** (about 0.7 V for silicon transistors, 0.3 V for germanium) before any appreciable base current flows at all. Once the base has been elevated to the bias voltage, any further increase in base voltage will start the base current flowing. This is shown in graph form in Figure 4.16. The **biasing circuit** of a transistor amplifier establishes a background “ballpark” DC voltage at the base. The signal to be amplified is then superimposed on this bias voltage. One common method of creating the bias voltage is with a resistor-type voltage-divider circuit as shown in Figure 4.17. Notice that resistors  $R_1$  and  $R_2$  create a base bias voltage of 0.8 V (ignoring any loading effects). Consulting the input curve of Figure 4.16, we see that 0.8 V would cause a base current  $I_B$  of 2 mA.

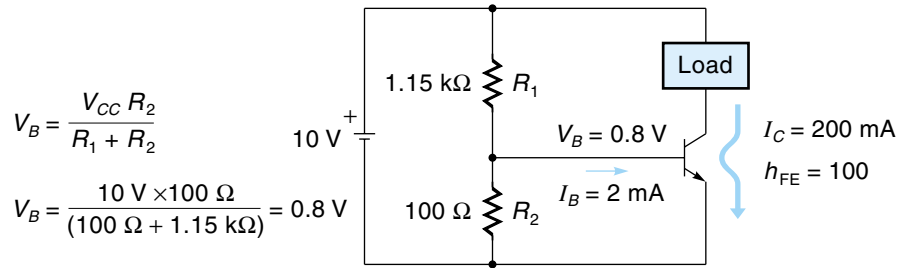
**Figure 4.16**

A silicon transistor input (base) curve.



**Figure 4.17**

A voltage-divider bias circuit.



The  $h_{FE}$  of the transistor is given as 100, so rearranging Equation 4.1 we can calculate the collector current  $I_C$ :

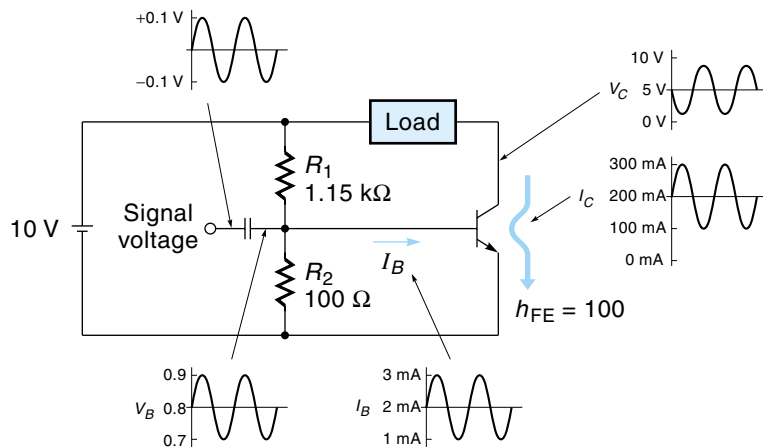
$$I_C = h_{FE} I_B = 100 \times 2 \text{ mA} = 200 \text{ mA}$$

Notice that the load is in series with the collector so that all 200 mA is going through the transistor and the load.

We are now ready to discuss the operation of the simple transistor power amplifier shown in Figure 4.18. The DC bias conditions have already been discussed in the preceding paragraph. Now consider an input signal voltage of  $\pm 0.1 \text{ V}$ , which is superimposed onto the base through a capacitor. In other words, the input is added to the 0.8 Vdc bias voltage to form an AC base voltage that cycles from 0.7 to 0.9 V, as shown in Figure 4.18. Referring again to the base curve (Figure 4.16), we see that a base voltage of 0.7-0.9 V translates into a base current of 1-3 mA, which the transistor multiplies into a collector current of 100-300 mA, and all of this current goes through the load. Notice that the output voltage  $V_C$  is out of phase with the output current  $I_C$ . This occurs because the transistor must lower its resistance to increase the current, and a lower resistance creates a lower voltage drop.

**Figure 4.18**

A simple transistor amplifier.



The circuit in Figure 4.18 is a class A amplifier. Specifically, **class A operation** is when the bias point is set somewhere in the middle of the operating range, and the actual output current is above and below this value. In other words, the transistor is biased to be “half-on” so that the input signal can turn it “more on” or “less on” from that middle point. Class A operation allows for both positive and negative input voltages to be amplified but has the drawback that power is being dissipated at all times, even when no input signal is present (because  $I_C$  is always flowing).

### EXAMPLE 4.2

Find the load current in the transistor circuit shown in Figure 4.19. Use the input curves in Figure 4.16 and assume that the  $h_{FE}$  of the transistor is 70.

### SOLUTION

We want to find the output current, and because we know the current gain is 70, the problem comes down to finding out what the base current is. In most cases, such as this one, it is the base voltage that we can calculate (or measure), so we will use the input curves to find the corresponding base current. First, calculate the base bias voltage from  $R_1$  and  $R_2$  using the voltage-divider rule:

$$V_{BE} = \frac{V_C R_2}{(R_1 + R_2)} = \frac{10 \text{ V} \times 100 \Omega}{1.2 \text{ k}\Omega + 100 \Omega} = 0.77 \text{ V}$$

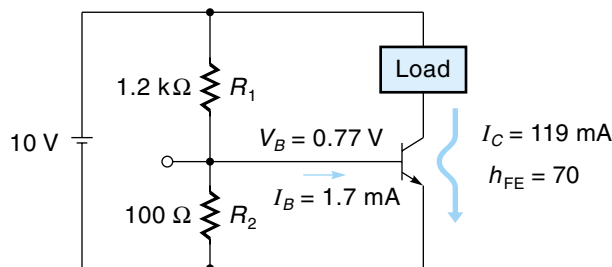
Now, knowing that the base voltage is 0.77 V, we use the input curve of Figure 4.16 to determine that  $I_B$  is about 1.7 mA. The load current  $I_C$  can now be calculated by rearranging Equation 4.1:

$$I_C = h_{FE} I_B = 70 \times 1.7 \text{ mA} = 119 \text{ mA}$$

*Note:* The purpose of this example was to give you insight on how the class A transistor actually works. Unfortunately, in most cases, the input curves are not presented in spec sheets, so transistor amplifier designers make use of feedback and various empirical relationships, which are presented in any good transistor design text.

**Figure 4.19**

A transistor circuit (Example 4.2).



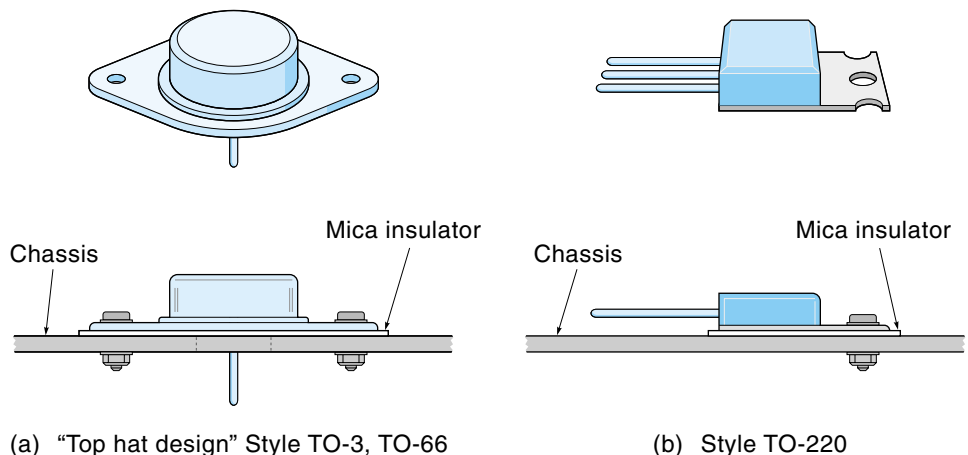
There are other biasing possibilities besides class A. In **class B operation**, the base bias voltage is set just under 0.7 V (for silicon), which is the point where the transistor is just at the brink of turning on. The advantage of this system is that no power is dissipated when there is no input signal (because the transistor is off). The disadvantage is that it can amplify only positive base voltages for NPN transistors (or negative base voltages for PNP).

The third biasing possibility is **class C operation** where the transistor is operated like a switch, either full-on or full-off. The base voltage will be either 0 V, which will turn it completely off, or high enough (say, 1 V) to turn it on so much that the transistor acts like a short circuit. The advantage of this system is that it is very power efficient (for those applications that can use the on-off type of power). Applications of class C amplifiers are discussed in Chapter 7.

Power transistors are physically bigger than signal transistors and are designed to carry large currents. In control systems, they are used to provide the drive current for motors and other electromechanical devices. When a transistor has a large current and voltage at the same time, the resulting power ( $V_C J_C$ ) must be dissipated in the form of heat. A typical power transistor is designed to operate up to 200°C (360°F) above ambient temperature. However, its power capacity is derated proportionally for temperatures above 25°C (as denoted by the spec sheet). The power transistor case has a flat metal surface to provide a thermal escape path for the heat. Therefore, to operate at anywhere near the rated power, the transistor must be mounted firmly to the chassis or a metal **heat sink**—a piece of metal with cooling fins to dissipate the heat into the air. Figure 4.20 illustrates two standard types of power transistors and how they are to be mounted. Many times the case itself is the collector terminal. If the collector must be kept electrically insulated from the mounting chassis, then a special mica insulator is used, together with a thermally conducting white grease.

**Figure 4.20**

Styles and mounting of power transistors.



**EXAMPLE 4.3**

A 100-watt (W) transistor fails, apparently from overheating. The transistor was replaced, and measurements showed that the load current ( $I_C$ ) was 2 A,  $V_{CE}$  was 35 V, and the case temperature rose to 125°C. What is the problem?

**SOLUTION**

First, calculate the power being dissipated by the transistor (using Equation 4.4):

$$P_D = I_C V_{CE} = 2 \text{ A} \times 35 \text{ V} = 70 \text{ W}$$

Seventy watts is certainly less than the manufacturer's designation of 100 W. The actual power capacity, however, will be less in this situation because the case is 100°C over ambient temperature (ambient is taken to be 25°C). Consulting a spec sheet, we find that the power must be derated 0.57 W/°C (above 25°). We now calculate the actual reduced power capacity:

$$P_D = 100 \text{ W} - (0.57 \text{ W/°C} \times 100\text{°C}) = 43 \text{ W}$$

Clearly, the problem is that 70 W exceeds the derated power capacity of 43 W. Apparently, the transistor is not “heat sunked” well enough; that is, the heat cannot conduct itself away from the transistor fast enough.

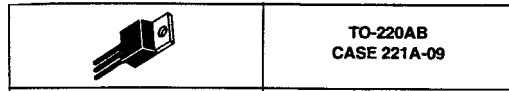
Figure 4.21 presents a sample selection sheet for some power transistors. For example, consider the TIP29C (the first NPN type on the list). This device can have a continuous load current ( $I_C$ ) of 1 A and a maximum  $V_{CE}$  of 100 V. Notice the wide range given for  $h_{FE}$  (15-75). It is common for transistors with the same part number to have widely different values of  $h_{FE}$ , so most transistor circuits are designed to accommodate this. You might notice that some transistors in Figure 4.21 have much higher values of  $h_{FE}$  than the rest. For example, the TIP112 (the fourth one down the list) has an  $h_{FE}$  value of 500 minimum. These are darlington transistors. As shown in the little drawing at the top of Figure 4.21, a **Darlington** actually has two transistors connected in such a way that there are still only three terminals presented to the outside world. It works like this: The input base current is amplified by the first transistor, and then *its entire output becomes the base current of the second transistor*, so the signal gets amplified a second time. If the  $h_{FE}$  of each individual transistor were 20, the combined  $h_{FE}$  would be  $20 \times 20 = 400$ .

Also, the gain of any transistor eventually decreases if the frequency gets high enough. The **gain-bandwidth product** ( $f_T$ ) is a constant that is the product of the frequency and the gain at that frequency. Therefore the gain-bandwidth product represents the frequency that is high enough to cause the gain ( $h_{FE}$ ) to reduce to 1.

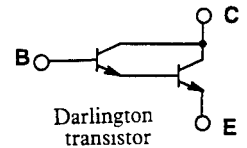
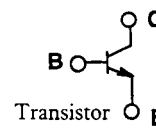
**Figure 4.21**

A selection sheet for power transistors. (Copyright © Semiconductor Components Industries, LLC. Used by permission.)

## BIPOLAR POWER TRANSISTORS SELECTOR GUIDE



TO-220AB  
CASE 221A-09



**Table 1. PLASTIC TO-220AB — CASE 221A-09, Style 1**

I <sub>C</sub> Cont Amps Max	V <sub>CE0(sus)</sub> Volts Min	Device Type		h <sub>FE</sub> Min/Max	β I <sub>C</sub> Amp	Resistive Switching			t <sub>r</sub> MHz Min	P <sub>D</sub> (Case) Watts @ 25°C
		NPN	PNP			t <sub>s</sub> μs Max	t <sub>f</sub> μs Max	@ I <sub>C</sub> Amp		
1	100	TIP29C	TIP30C	15/75	1	0.6 typ	0.3 typ	1	3	30
	250	TIP47		30/150	0.3	2 typ	0.18 typ	0.3	10	40
	400	<b>TIP50</b>	<b>MJE5731A</b>	30/150	0.3	2 typ	0.18 typ	0.3	10	40
2	100	<b>TIP112(2)</b>	<b>TIP117(2)</b>	500 min	2	1.7 typ	1.3 typ	2	25 <sup>(1)</sup>	50
	450/1000	<b>BUX85</b>		30	0.1	3.5	1.4	1	4	50
	450/1000	<b>MJE18002</b>		14/34	0.2	3	0.17	1	12 typ	40
3	100	<b>TIP31C</b>	<b>TIP32C</b>	25 min	1	0.6 typ	0.3 typ	1	3	40
4	80	<b>D44C12</b>	<b>D45C12</b>	40/120	0.2			1	40 typ	30
	400/700	<b>MJE13005</b>		6/30	3	3	0.7	3	4	60
5	100	<b>TIP122(2)</b>	<b>TIP127(2)</b>	1k min	3	1.5 typ	1.5 typ	4	4 <sup>(1)</sup>	75
	400/700	<b>BUL45</b>		14/34	0.3	1.7	0.15	1	12 typ	75
	450/1000	MJE16002		5 min	5	3	0.3	3		80
	550/1200	<b>MJE18204</b>		18/35	0.5	2.75	0.2	2	12	75
6	100	<b>TIP41C</b>	<b>TIP42C</b>	15/75	3	0.4 typ	0.15 typ	3	3	65
	400/700	<b>BUL146</b>		14/34	0.5	1.75	0.15	3	14 typ	100
7	70	<b>2N6292</b>	<b>2N6107</b>	30/150	2	0.4 typ	0.15 typ	3	4	40
	150	BU407		30 min	1.5		0.75	5	10	60
8	100	<b>BDX53C(2)</b>	<b>BDX54C(2)</b>	750 min	3					
		<b>TIP102(2)</b>	<b>TIP107(2)</b>	1k/20k	3	1.5 typ	1.5 typ	3	4 <sup>(1)</sup>	80
	150	<b>MJE15030</b>	<b>MJE15031</b>	20 min	4				30	50
	250	<b>MJE15032</b>	<b>MJE15033</b>	50 min	1				30	50
	350	<b>MJE13007</b>		5/30	5	3	0.7	5		80
			<b>MJE5852</b>	15 min	2	2	0.5	4		80
400/700	<b>BUL147</b>		14/34	1	2.5	0.18	2	14 typ	125	
10	60	<b>MJE3055T</b>	<b>MJE2955T</b>	20/70	4					75
	80	<b>2N6388(2)</b>	<b>2N6668(2)</b>	1k/20k	5				20 <sup>(1)</sup>	65
		<b>D44H11</b>	<b>D45H11</b>	40 min	4	0.5 typ	0.14 typ	5	50 typ	50
	100	<b>BDX33C(2)</b>	<b>BDX34C(2)</b>	750 min	3				3	70
12	400/700	<b>MJE13009</b>		6/30	8	3	0.7	8	4	100
15	80	<b>D44VH10</b>	<b>D45VH10</b>	20 min	4	0.5	0.09	8	50 typ	83
	100	<b>BDW42(2)</b>	<b>BDW47(2)</b>	1k min	5	1 typ	1.5 typ	5	4	85

<sup>(1)</sup> |h<sub>FE</sub>| @ 1 MHz

<sup>(2)</sup> Darlington

Devices listed in bold, italic are ON Semiconductor preferred devices.



**EXAMPLE 4.4**

A transistor in a piece of equipment has gone bad and needs to be replaced with one of those listed in Figure 4.21. The specifications for the original transistor (now bad) are the following:

Type: NPN

Maximum load current: 8 A

Maximum voltage across the transistor: 150 V

Maximum power dissipation: 50 W

$h_{FE}$ : 30

**SOLUTION**

First consider the load current of 8 A. Looking down the list of 8-amp NPN transistors, we see the MJE15030, which meets the  $V_{CEO}$  requirement (150 V) but not the  $h_{FE}$ . Thus we will have to go with MJE15032 ( $h_{FE} = 50$  min), which meets all the requirements.

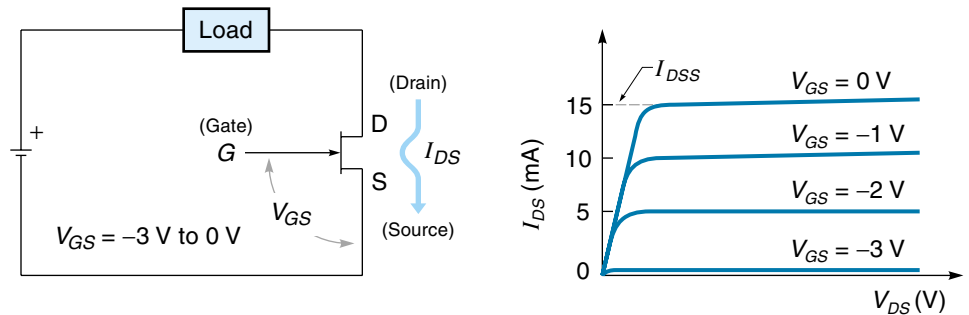
**Field Effect Transistors (FET)**

The **field effect transistor** (FET) is another three-terminal solid-state amplifying device. FETs perform the same job as the BJTs of the previous section, but because of some advantages are becoming more common in power applications. These advantages include high input impedance, high switching speeds, and less temperature sensitivity. There are two types of FETs, the **junction FET** (JFET) and the **metal-oxide semiconductor FET** (MOSFET). Both types have three terminals; the **drain** (D), the **source** (S), and the **gate** (G). Also, FETs are made to be either N-channel or P-channel; the type determines the direction of current through the device. The FETs discussed in this section are all N-channel.

The simple JFET circuit is shown in Figure 4.22(a). The load current  $I_{DS}$  passes through the drain (D) to the source (S). The amount of drain current is controlled by the voltage between the gate and the source ( $V_{GS}$ ) (as contrasted with the BJT, where the collector current is controlled by the base current). The characteristics of the JFET are described by the characteristic curves shown in Figure 4.22(b). Notice that the load current ( $I_{DS}$ ) is on the vertical axis, and the voltage across the FET ( $V_{DS}$ ) is on the horizontal axis. Each curve represents what happens for a different gate voltage. For example, when the gate voltage ( $V_{GS}$ ) is  $-3$  V, the  $I_{DS}$  will be 0 mA. This gate voltage is known as  $V_{GS(off)}$ . When the gate voltage is  $-2$  V,  $I_{DS}$  will be 5 mA, and so on, until the gate voltage is 0 V, causing  $I_{DS}$  to be 15 mA. Because the gate voltage is not allowed to go positive in an N-channel JFET, 15 mA is the highest  $I_{DS}$  for this particular FET and is called  $I_{DSS}$  (drain-source current with the gate shorted).

**Figure 4.22**

Junction field effect transistor (JFET) operation.



(a) Simple JFET circuit

(b) FET characteristic curves (sample)

The second type of FET is the MOSFET. MOSFETs are particularly important in this text because *power MOSFETs*, which can handle many amperes, are becoming more popular. In Figure 4.23(a), notice that in the FET symbol the gate is shown as not even touching the rest of the device. This means that the gate is capacitively coupled, causing a very high DC input resistance and allowing for what is called the **enhancement mode** of operation. In the enhancement mode, all gate voltages are positive. Consider the sample characteristic curves of a power-type MOSFET shown in Figure 4.23(b). When the  $V_{GS}$  is 4 V, the  $I_{DS}$  is 0 A. When the  $V_{GS}$  is 5 V, the  $I_{DS}$  is 1 A; when the  $V_{GS}$  is 6 V, the  $I_{DS}$  is 2 A (and so on, up to the maximum current allowed by the device).

For both the JFET and the MOSFET, the input is the gate voltage and the output is the drain-source current. The gain of this device is called the **transconductance** ( $g_m$ ) and is expressed in Equation 4.5:

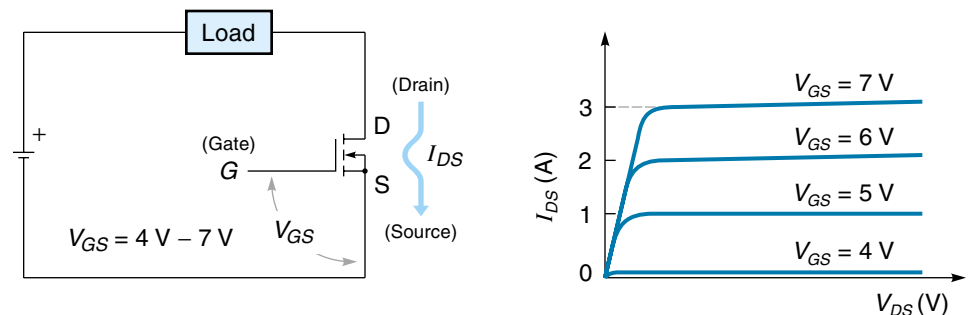
$$\text{Gain} = \frac{\Delta \text{output current}}{\Delta \text{input voltage}} = g_m = \frac{\Delta I_{DS}}{\Delta V_{GS}} S \text{ (or mho)} \quad (4.5)$$

where

$g_m$  = FET gain, called the transconductance

**Figure 4.23**

A metal oxide semiconductor field effect transistor (MOSFET).



(a) Simple MOSFET circuit

(b) MOSFET characteristic curves (sample)

$$I_{DS} = \text{load current (from drain to source)}$$

$$V_{GS} = \text{control voltage at the gate}$$

Notice that the unit of  $g_m$  is  $1/R$ , known as *siemens* (S). An older term for conductance (still in use) is *mho*, which is *ohm* spelled backwards.

### EXAMPLE 4.5

For the power MOSFET whose characteristic curves are shown in Figure 4.23(b), find the gain  $g_m$ .

### SOLUTION

From Equation 4.5, we see that the gain  $g_m$  is the *change in output current due to the corresponding change in input gate voltage*. Thus, we pick two convenient gate voltages—say, 5 V and 6 V (which represents a change of 1 V)—and divide that into the corresponding change in the load current  $I_{DS}$ :

$$\text{Gain} = g_m = \frac{\Delta I_{DS}}{\Delta V_{GS}} = \frac{(2 \text{ A} - 1 \text{ A})}{(6 \text{ V} - 5 \text{ V})} = \frac{1 \text{ A}}{1 \text{ V}} = 1 \text{ S (or 1 mho)}$$

The specifications for the Motorola MTW24N40E power MOSFET are shown in Figure 4.24. Some important characteristics of this MOSFET are as follows:

1. Maximum load current ( $I_{DS}$ ) is 24 A.
2. The gate threshold voltage ( $V_{GS(th)}$ ), the gate voltage that just causes the FET to conduct, is between 2 and 4 V.
3. The minimum FET gain, called the forward transconductance ( $g_{FS}$ ), is 11 mho. (Note:  $g_{FS}$  is the same as  $g_m$ .)
4. Total power dissipation for the FET ( $P_D$ ) should be less than 250 W.

### EXAMPLE 4.6

A particular power FET (MTW24N40E) was found to just start conducting when  $V_{GS} = 2.5 \text{ V}$ . Also, when  $V_{GS} = 3 \text{ V}$ , the load current ( $I_D$ ) was measured to be 6 A. Is this FET operating within specifications?

### SOLUTION

The spec sheet shows that the gate-threshold voltage  $V_{GS(th)}$  can be between 2 V and 4 V. Thus if this FET starts conducting with a gate voltage of 2.5 V, it meets this spec.

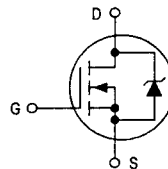
**Figure 4.24**

Specification for the MTW24N40E power MOSFET. (Copyright © Semiconductor Components Industries, LLC. Used by permission.)

*Designer's™ Data Sheet*  
**TMOS E-FET™**  
**Power Field Effect Transistor**  
**TO-247 with Isolated Mounting Hole**  
**N-Channel Enhancement-Mode Silicon Gate**

This high voltage MOSFET uses an advanced termination scheme to provide enhanced voltage-blocking capability without degrading performance over time. In addition, this advanced TMOS E-FET is designed to withstand high energy in the avalanche and commutation modes. The new energy efficient design also offers a drain-to-source diode with a fast recovery time. Designed for high voltage, high speed switching applications in power supplies, converters and PWM motor controls, these devices are particularly well suited for bridge circuits where diode speed and commutating safe operating areas are critical and offer additional safety margin against unexpected voltage transients.

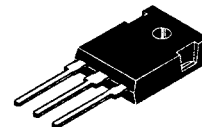
- Robust High Voltage Termination
- Avalanche Energy Specified
- Source-to-Drain Diode Recovery Time Comparable to a Discrete Fast Recovery Diode
- Diode is Characterized for Use in Bridge Circuits



**MTW24N40E**

Motorola Preferred Device

TMOS POWER FET  
 24 AMPERES  
 400 VOLTS  
 $R_{DS(on)} = 0.16 \text{ OHM}$



CASE 340K-01, Style 1  
 TO-247AE

**MAXIMUM RATINGS** ( $T_C = 25^\circ\text{C}$  unless otherwise noted)

Rating	Symbol	Value	Unit
Drain-Source Voltage	$V_{DSS}$	400	Vdc
Drain-Gate Voltage ( $R_{GS} = 1.0 \text{ M}\Omega$ )	$V_{DGR}$	400	Vdc
Gate-Source Voltage — Continuous	$V_{GS}$	$\pm 20$	Vdc
— Non-Repetitive ( $t_p \leq 10 \text{ ms}$ )	$V_{GSM}$	$\pm 40$	Vpk
Drain Current — Continuous	$I_D$	24	Adc
— Continuous @ $100^\circ\text{C}$	$I_D$	17.7	
— Single Pulse ( $t_p \leq 10 \mu\text{s}$ )	$I_{DM}$	72	Apk
Total Power Dissipation	$P_D$	250	Watts
Derate above $25^\circ\text{C}$		2.0	W/ $^\circ\text{C}$

Characteristic	Symbol	Min	Typ	Max	Unit
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**OFF CHARACTERISTICS**

Drain-Source Breakdown Voltage ( $V_{GS} = 0 \text{ Vdc}$ , $I_D = 250 \mu\text{Adc}$ ) Temperature Coefficient (Positive)	$V_{(BR)DSS}$	400	—	—	Vdc
		—	360	—	mV/ $^\circ\text{C}$
Zero Gate Voltage Drain Current ( $V_{DS} = 400 \text{ Vdc}$ , $V_{GS} = 0 \text{ Vdc}$ ) ( $V_{DS} = 400 \text{ Vdc}$ , $V_{GS} = 0 \text{ Vdc}$ , $T_J = 125^\circ\text{C}$ )	$I_{DSS}$	—	—	10	$\mu\text{Adc}$
		—	—	100	
Gate-Body Leakage Current ( $V_{GS} = \pm 20 \text{ Vdc}$ , $V_{DS} = 0$ )	$I_{GSS}$	—	—	100	nAdc

**ON CHARACTERISTICS (1)**

Gate Threshold Voltage ( $V_{DS} = V_{GS}$ , $I_D = 250 \mu\text{Adc}$ ) Temperature Coefficient (Negative)	$V_{GS(th)}$	2.0	—	4.0	Vdc
		—	7.0	—	mV/ $^\circ\text{C}$
Static Drain-Source On-Resistance ( $V_{GS} = 10 \text{ Vdc}$ , $I_D = 12 \text{ Adc}$ )	$R_{DS(on)}$	—	0.13	0.16	Ohm
Drain-Source On-Voltage ( $V_{GS} = 10 \text{ Vdc}$ ) ( $I_D = 24 \text{ Adc}$ ) ( $I_D = 12 \text{ Adc}$ , $T_J = 125^\circ\text{C}$ )	$V_{DS(on)}$	—	—	4.5	Vdc
		—	—	4.3	
Forward Transconductance ( $V_{DS} = 15 \text{ Vdc}$ , $I_D = 12 \text{ Adc}$ )	$g_{FS}$	11	17	—	mhos

The second set of data gives one operating point. Specifically, we are told that an input voltage ( $V_{GS} = 3 \text{ V}$ ) gives a corresponding output current ( $I_D = 6 \text{ A}$ ). These numbers allow us to calculate the gain (transconductance)

$$\text{Gain} = g_m = g_{FS} = \frac{\Delta I_D}{\Delta V_{GS}} = \frac{6 \text{ A} - 0 \text{ A}}{3 \text{ V} - 2.5 \text{ V}} = \frac{6 \text{ A}}{0.5 \text{ V}} = 12$$

The operating value of  $g_{FS} = 12$  is above the minimum spec sheet value of 11, so the FET is OK.

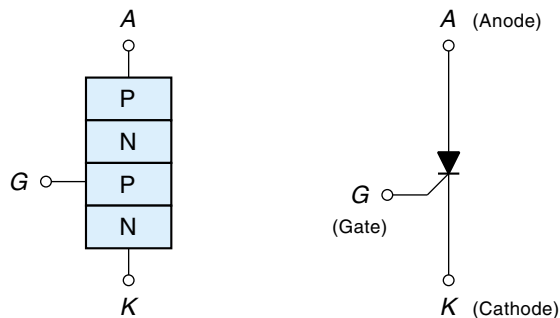
## 4.4 SILICON-CONTROLLED RECTIFIERS

The **silicon-controlled rectifier** (SCR) is a three-terminal power-control device that is a member of a group known as **thyristors**. Thyristors are four-layer semiconductor devices that behave like a switch; that is, they are either conducting or nonconducting (on or off). Constructed from four layers of semiconductor material (PNPN), as shown in Figure 4.25(a), the SCR has three terminals: the **anode** (A), the **cathode** (K), and the **gate** (G), as shown in the schematic symbol [Figure 4.25(b)].

The SCR is usually connected in series with the load as shown in Figure 4.26. Notice that this circuit is similar to a transistor or FET power-driving circuit. In this case, the load current enters the anode and leaves through the cathode. The gate is used to switch the SCR from its nonconduction state (off) to its conduction state (on). Unfortunately *the gate cannot be used to turn the SCR off*, as will be addressed later in this section. Also, current can travel in only one direction through the SCR (as it does in a diode).

The best way to explain the operation of the SCR is with the characteristic curve, which is given in Figure 4.27. Take time to consider this graph. First recognize that voltage across the SCR is on the horizontal axis, and current through the SCR is on the vertical axis. The left side of the graph corresponds to the SCR being reversed-biased,

**Figure 4.25**  
A silicon-controlled rectifier.

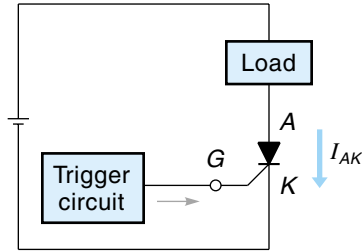


(a) SCR construction

(b) SCR schematic symbol

**Figure 4.26**

The basic SCR circuit.



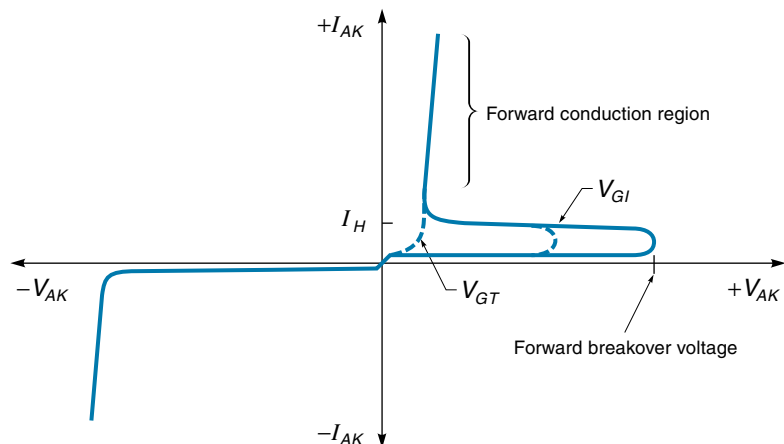
which explains why there is no current flow (until reverse breakdown occurs at some high negative voltage). The interesting action takes place on the right half of the graph, where the *solid line shows the action of the SCR when there is no gate voltage*. As the voltage across the SCR gets more positive, the current is held off until the voltage reaches a value known as the **forward breakover voltage**, at which point the SCR turns on and current starts to flow (and the blocking voltage shrinks back). The SCR is now in the **forward conduction region** and is acting like a low-resistance closed switch. If the current is now reduced below the value known as the **holding current** ( $I_H$ ), the SCR will immediately switch back into its high-resistance nonconducting state.

We now turn our attention to how the gate voltage affects the operation of the SCR. When the gate voltage is  $V_{G1}$  (in Figure 4.27), the forward breakover voltage is significantly reduced—that is, the SCR will turn on at a lower anode-cathode voltage (indicated by the dashed line). If the gate voltage is high enough (called  $V_{GT}$ ), the SCR starts conducting right away and requires almost no forward breakover voltage whatsoever. This is the turn-on mode most practical SCR circuits use. Once the SCR starts to conduct, the gate voltage can be removed, and the SCR will remain in the conduction state. Therefore, in many SCR circuits, the gate signal is actually a short “trigger” pulse, just long enough to ensure that switching has occurred.

Turning on the SCR with the gate is relatively easy, but turning it off is another matter. *There is no way to command the SCR to turn off*; the only way to turn it off is

**Figure 4.27**

An SCR characteristic curve (sample).



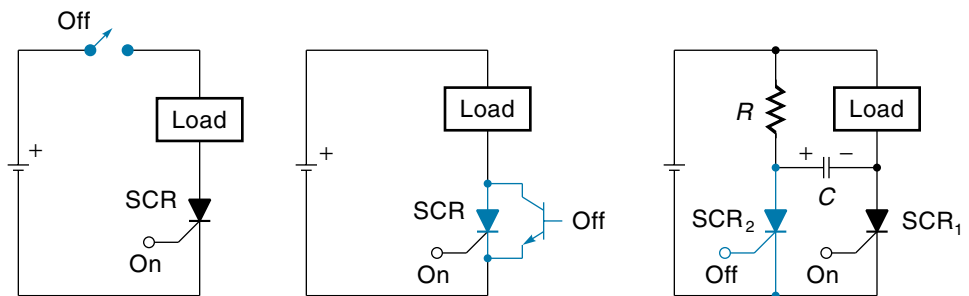
to reduce the load current below the holding current ( $I_H$ ) value. For DC circuits, this can be done in a number of ways. The first method, illustrated in Figure 4.28(a), is to open the load circuit with some other device—say, a switch. The second method, called **forced commutation**, involves momentarily forcing the voltage across the SCR ( $V_{AK}$ ) to zero (or below), thereby stopping the current. One way to do this is illustrated in Figure 4.28(b), where a “shorting transistor” momentarily reroutes the current around the SCR. Another way is shown in Figure 4.28(c) and uses a second SCR<sub>2</sub> to turn off the main SCR<sub>1</sub>. The circuit works as follows: When the main SCR<sub>1</sub> is on, its anode voltage is practically 0 V; therefore, capacitor  $C$  can charge up through resistor  $R$ . To turn off the main SCR<sub>1</sub>, SCR<sub>2</sub> is triggered into conduction which “pulls down” the voltage at the positive (+) end of  $C$  to 0 V, temporarily causing the negative (–) end of  $C$  to drop as well. The negative voltage on the anode of SCR<sub>1</sub> turns it off. Finally, when the capacitor discharging current wanes, SCR<sub>2</sub> turns off by itself.

Turning off the SCR for AC circuits is not a problem because it will naturally turn itself off at each negative half-cycle (remember, the SCR can only conduct in one direction). So for AC applications, the SCR must be triggered on once for each positive half-cycle, and then it will turn itself off at the end of that positive half-cycle.

Because SCRs work so well with AC, they are often used as the rectifier in an AC-to-DC converter—for example, to control the speed of a DC motor. A simple SCR speed-control circuit is shown in Figure 4.29(a). As you can see, the gate voltage is supplied by the AC itself through potentiometer  $R$  and diode  $D$ . When the level of the gate voltage gets high enough, the SCR turns on, and it stays on for the remainder of that positive half-cycle. By increasing the resistance of  $R$ , the time at which the SCR fires can be delayed, thus causing a shorter conduction period. These power bursts are averaged out by the load (such as a motor) into an effective DC voltage.

The circuit just described can only control the trigger time from 0 to 90° of the sine wave. To delay the trigger point beyond 90°, a capacitor can be added to the gate circuit as shown in Figure 4.29(b). As the capacitor  $C$  charges up through  $R$ ,  $V_C$  lags behind (remember, this happens for each cycle). Using this circuit, known as a **half-wave phase-control circuit**, the SCR firing can be delayed up to about 170°, resulting in such a narrow “on time” that the DC equivalent is practically 0 V.

**Figure 4.28**  
Turning off an SCR  
for a DC load.



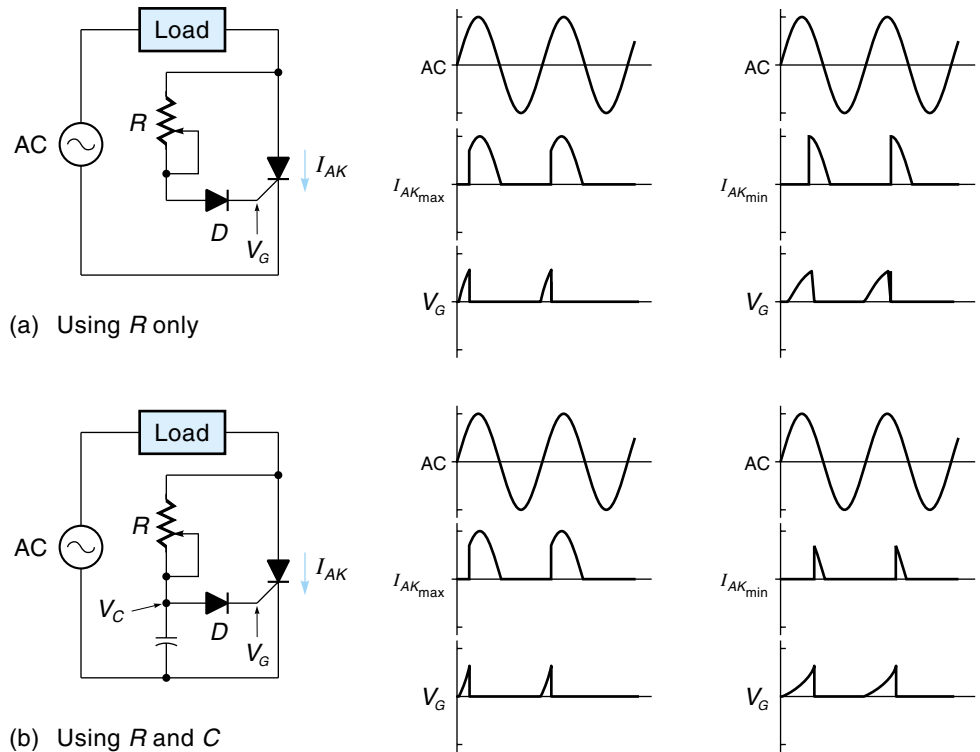
(a) Using a switch

(b) Using a transistor

(c) Using a second SCR

**Figure 4.29**

Turning on an SCR for an AC load. (a) Trigger circuit uses a resistor only and (b) trigger circuit uses a resistor and capacitor.



The AC-to-DC converter circuits described in the last two paragraphs are half-wave rectifiers, which means only half of the potential AC power is used. The SCR can be used in conjunction with a full-wave rectifier to make use of more of the AC power by giving two power pulses per cycle. Such a circuit is shown in Figure 4.30 and is known as a *full-wave phase-control circuit*. The waveform (Figure 4.30) for the SCR current ( $I_{AK}$ ) depicts the situation when conduction starts at about  $90^\circ$ . On real systems, voltage is easier to measure than current, so it is important to know what the voltage waveforms look like. Notice that during conduction the SCR voltage ( $V_{AK}$ ) is practically zero, and when the SCR is off, it assumes the entire line voltage.

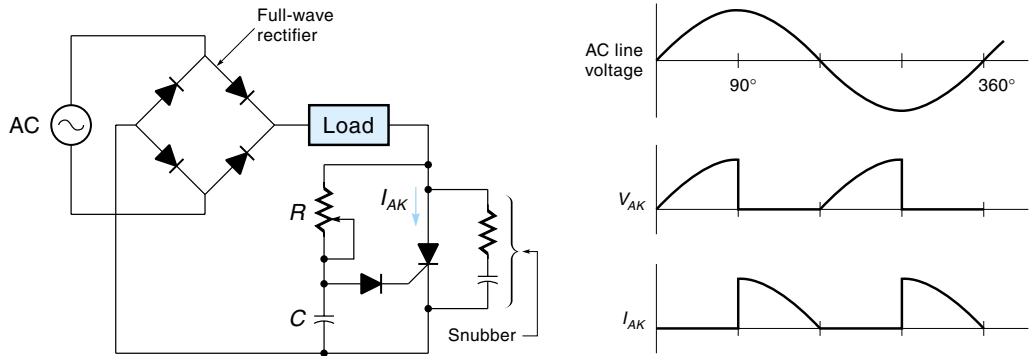
Some other conditions (besides those already mentioned) could cause the SCR to turn on:

1. If the voltage across the SCR ( $V_{AK}$ ) rises too quickly, the SCR may turn on, even if the voltage is below the forward breakover voltage. This is called the **dv/dt effect**. To eliminate this possibility,  $R_C$  **snubber** circuits are connected across the SCR to prevent the voltage from rising too fast (see Figure 4.30).
2. After turning off,  $V_{AK}$  and  $I_H$  must be held low for a given period of time or the SCR will turn on again.
3. As the SCR temperature increases, the required breakover voltage decreases, thus possibly causing the SCR to turn on.



**Figure 4.30**

An SCR used with a full-wave rectifier.



4. A different type of SCR, known as a LASCR (light-activated SCR), uses light energy to trigger the conduction state. These devices have a little window that allows the light to hit the necessary PN junction.

SCRs are available in a wide range of sizes, with current capacities from 0.5 to over 1000 A. A specification sheet of some Motorola SCRs is given in Figure 4.31. An explanation of the parameters used is as follows:

- $V_{DRM}$ : peak forward blocking voltage
- $V_{RRM}$ : peak reverse blocking voltage
- $I_{TSM}$ : peak surge current
- $I_{GT}$ : gate trigger current, continuous DC
- $V_{GT}$ : gate trigger voltage, continuous DC

For example, a 2N6401 (Figure 4.31, seventh column, second row) can carry 16 A of load current, should be used with supply voltages under  $100 V_{\text{peak}}$  (+ or -), can handle brief surge load currents of up to 160 A, and requires 30 mA of gate current to turn on, which can be achieved with 1.5 V on the gate.

#### EXAMPLE 4.7

A 15-A 120-Vdc motor is to be powered from a full-wave SCR circuit similar to that shown in Figure 4.30. The source of power is 120 Vac. Select an SCR to do the job from the list given in Figure 4.31.

#### SOLUTION

Although the motor is DC, it is getting power in the form of pulses of rectified AC. We can calculate the peak voltage as follows:

$$V_{\text{peak}} = \frac{120 \text{ V}}{0.707} = 170 \text{ V}$$

Fifteen amps dc is equivalent to  $15 A_{\text{RMS}}$ , so given the choices available, we select a 16-A unit that can take 400 V. A 2N6403 would be a good choice.

**Figure 4.31**

Specification sheet for SCRs. (Copyright © Semiconductor Components Industries, LLC. Used by permission.)

**ON Semiconductor Selector Guide – Discrete Devices**

**Table 1. SCRs — General Purpose Plastic Packages (continued)**

		On-State (RMS) Current									
		10 AMPS		12 AMPS				16 AMPS		25 AMPS	
		T <sub>C</sub> = 75°C	T <sub>C</sub> = 90°C	T <sub>C</sub> = 75°C	T <sub>C</sub> = 90°C	T <sub>C</sub> = 80°C		T <sub>C</sub> = 90°C	T <sub>C</sub> = 80°C	T <sub>C</sub> = 85°C	
V <sub>DRM</sub> V <sub>RRM</sub> (Volts)											
		Sensitive Gate		Sensitive Gate		High Performance		High Performance			
		Case 221A-07 TO-220AB Style 3	DPAK Case 369A-13 Style 4		Case 221A-07 TO-220AB Style 3	Case 221A-09 TO-220AB Style 3		Case 221A-07 TO-220AB Style 3	Case 221A-09 TO-220AB Style 3	Case 221A-07 TO-220AB Style 3	
50		—	—	—	2N6394	—	—	2N6400	—	2N6504	MCR69-2
100		—	—	—	2N6395	—	—	2N6401	—	2N6505	MCR69-3
200		—	—	—	—	—	—	—	—	—	—
400		MCR12LD <sup>(2)</sup>	—	—	2N6397	MCR12D <sup>(2)</sup>	—	2N6403	MCR25D <sup>(2)</sup>	2N6507	—
600		MCR12LM <sup>(2)</sup>	MCR12DCM <sup>(4)</sup>	MCR12DSM <sup>(4)</sup>	2N6398	MCR12M <sup>(2)</sup>	—	2N6404	MCR25M <sup>(2)</sup>	2N6508	—
800		MCR12LN <sup>(2)</sup>	MCR12DCN <sup>(4)</sup>	MCR12DSN <sup>(4)</sup>	2N6399	MCR12N <sup>(2)</sup>	MCR16N <sup>(2)</sup>	2N6405	MCR25N <sup>(2)</sup>	2N6509	—
<b>Maximum Electrical Characteristics</b>											
I <sub>TSM</sub> (Amps) 60 Hz	100					150	160	300		750 <sup>(7)</sup>	
I <sub>GT</sub> (mA)	8	20		30		20		30	30	40	30
I <sub>GT</sub> (μA)				200							
V <sub>GT</sub> (V)	1.5	1			1.5	2.2	1.7	1.5	1	1.5	
dV/dt V/μsec	Min.	Min.	Typical	Min.	Typical	Min.	Min.	Min.			
	50	50	200	2	10	50	50	50			
T <sub>J</sub> Op Range (°C)	-40 to +100		-40 to +125		-40 to +110		-40 to +125				

(2) No Suffix: Shipped in rails

(4) T4 Suffix: 2500/reel

(5) Add -001 suffix for straight lead DPAK

(7) Peak capacitor discharge current for t<sub>w</sub> = 1 ms. t<sub>w</sub> is defined as five time constants of an exponentially decaying current pulse (crowbar applications).

## 4.5 TRIACS

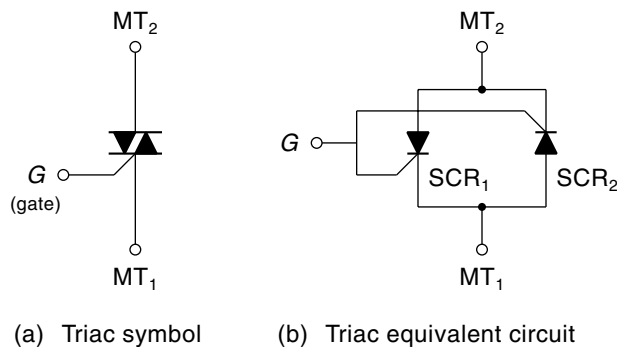
The **triac** is a three-terminal device that is similar to the SCR except that the triac can conduct current in both directions. Its primary use is to control power to AC loads such as turning AC motors on and off or varying the power for lighting and heating systems. As indicated on schematic symbol [Figure 4.32(a)], the three terminals are  $MT_1$ ,  $MT_2$ \* (main terminals), and the **gate**. As the name suggests, the load current passes through the main terminals, and the gate controls the flow. Figure 4.32(b) shows the equivalent circuit of the triac, which consists of two back-to-back SCRs with a common gate. When  $MT_2$  is more positive, the current flows through  $SCR_1$ ; when  $MT_1$  is more positive, the current flows through  $SCR_2$ .

The operating characteristics of the triac are best explained using the characteristic curve shown in Figure 4.33. Notice that the right half of the curve (in quadrant I) looks just like the SCR curve; no current flows until either the breakover voltage is reached or the gate is triggered (indicated by dashed line). This same pattern is repeated in Quadrant III (for voltage and current of the opposite direction). Also, like the SCR, once the triac is triggered on, it will remain on by itself until the load current drops below the holding current value ( $I_H$ ).

A single cycle of AC has a positive and a negative half-cycle. *The triac requires a trigger pulse at the gate for each half-cycle* and works best if the trigger is positive for the positive half-cycle and negative for the negative half-cycle (although in most cases the triac will also trigger if the gate goes negative in the positive half-cycle and if it goes positive in the negative half-cycle.).

A triac can be used as an on-off solid-state switch for AC loads or to regulate power to an AC load, such as a dimmer switch. A typical dimmer switch circuit is shown in Figure 4.34. The action of this circuit is similar to that of the SCR circuit in Figure 4.29(b). For each half-cycle, the capacitor  $C$  starts to charge up through resistor  $R$ . When the voltage  $V_C$  gets large enough, the triac triggers into conduction and stays on for the

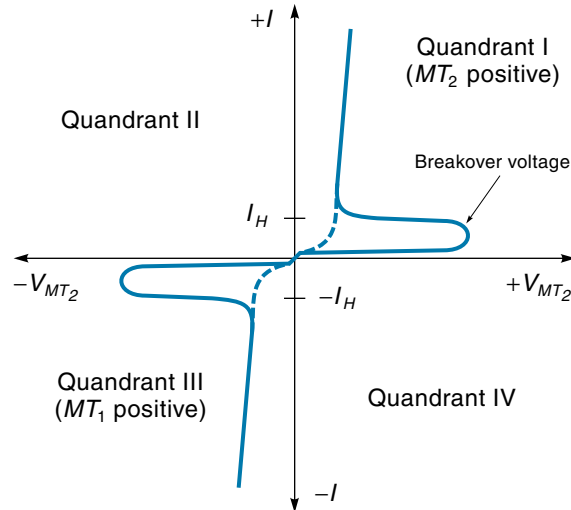
**Figure 4.32**  
Triac



\*Sometimes called *anode 1* and *anode 2*.

**Figure 4.33**

A triac characteristic curve (sample).



remainder of the half-cycle. By changing the resistance in  $R$ , the trigger point can be delayed, thus reducing the power to the load.

Triacs are available in various packages, some of which can handle currents up to 50 A (which is considerably less than the SCR). Figure 4.35 shows a sample selection of triacs. An explanation of the specifications follows.

$V_{\text{DRM}}$ : maximum off-state voltage allowed

$I_{\text{TSM}}$ : maximum load surge current

$I_{\text{GT}}$  and  $V_{\text{GT}}$ : gate current and voltage required to trigger triac into conduction

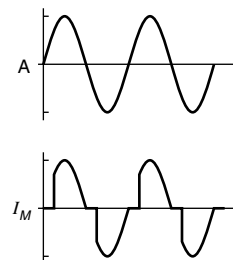
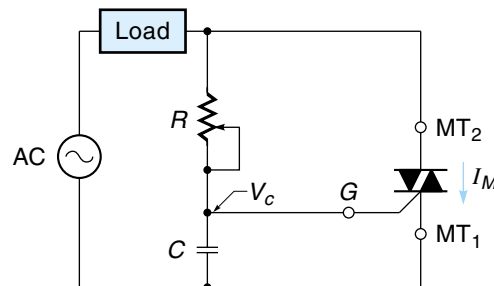
The two most common trigger conditions are  $\text{MT2}(+)\text{G}(+)$  and  $\text{MT2}(-)\text{G}(-)$ , as used in the circuit of Figure 4.34.

### Calculation of Delay and Conduction Periods

As we have seen in the preceding sections, phase-control circuits (using SCRs and triacs) regulate power to the load by turning on for a portion of the cycle period. As shown

**Figure 4.34**

A phase-control circuit using a triac (as might be used for a dimmer switch).

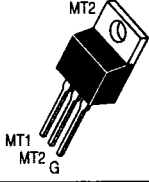


ON Semiconductor Selector Guide – Discrete Devices

Figure 4.35

A selection of triacs. (Copyright © Semiconductor Components Industries, LLC. Used by permission.)

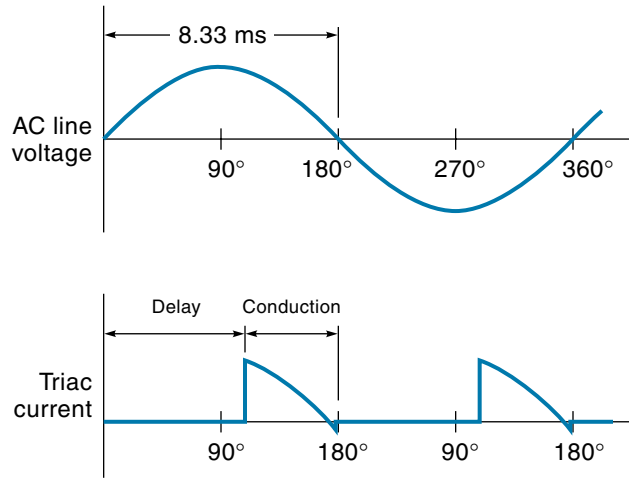
Table 2. TRIACs (continued)

V <sub>DRM</sub> (Volts)	On-State (RMS) Current		
	10 AMPS	12 AMPS	
	T <sub>C</sub> = 70°C	T <sub>C</sub> = 80°C	T <sub>C</sub> = 85°C
	 <p>Case 221A-07 TO-220AB Style 4</p>		
400	—	MAC12HCD(2)	—
600	MAC210A8	MAC12HCM(2)	MAC212A8
800	—	MAC12HCN(2)	MAC212A10
Maximum Electrical Characteristics			
I <sub>TSM</sub> (Amps)	100		
I <sub>GT</sub> @ 25°C (mA)			
MT2(+) G(+)	50	50	50
MT2(+) G(-)	50	50	50
MT2(-) G(-)	50	50	50
MT2(-) G(+)	75	—	75
V <sub>GT</sub> @ 25°C (V)			
MT2(+) G(+)	2	1.5	2
MT2(+) G(-)	2	1.5	2
MT2(-) G(-)	2	1.5	2
MT2(-) G(+)	2.5	—	2.5
T <sub>J</sub> Operating Range (°C)	-40 to +125		

(2) No Suffix: Shipped in rails

**Figure 4.36**

Delay and conduction periods in SCR and triac waveforms.



in Figure 4.36, for each half-cycle, the SCR (or triac) delays for a period of time and then switches into conduction for the rest of the half-cycle. Delay and conduction measurements are commonly expressed in units of time or as an angle. It is useful to be able to make conversions between these two measurements.

In North America, AC power has a frequency of 60 Hz. Therefore, the time for each half-cycle is

$$0.5 \times \frac{1}{60 \text{ Hz}} = 8.33 \text{ ms}$$

Therefore,

$$\text{Delay time} + \text{conduction time} = 8.33 \text{ ms} \quad (4.6)$$

So, for example, if an SCR had a conduction time of 3 ms, the delay time would be calculated as follows: 8.33 ms – 3 ms = 5.33 ms.

When dealing with angles, recall that each half-cycle is 180°. Therefore,

$$\text{Delay angle} + \text{conduction angle} = 180^\circ \quad (4.7)$$

So, for example, if a triac had a delay angle of 30°, then

$$\text{Conduction angle} = 180^\circ - 30^\circ = 150^\circ$$

Converting between time and angle measurements requires knowing the relationship between time and degrees. For a 60-Hz system, there are 8.33 ms per half-cycle of 180°. Therefore,

$$\frac{8.33 \text{ ms}}{180^\circ} = 46.3 \text{ } \mu\text{s/deg}$$

$$\text{or} \quad 46.3 \text{ } \mu\text{s} = 1 \text{ deg (for a 60-Hz system)} \quad (4.8)$$

So, for example, if a triac had a conduction angle of  $120^\circ$ , we could calculate the conduction time as follows:  $46.3 \mu\text{s}/\text{deg} \times 120^\circ = 5.56 \text{ ms}$ .

#### EXAMPLE 4.7

For a 60-Hz system, a particular SCR delays 5 ms before firing.

- (a) Find the conduction time for this SCR.
- (b) Find the conduction angle for this SCR.

#### SOLUTION

- (a) To find the conduction time, we apply Equation 4.6:

$$\text{Delay time} + \text{conduction time} = 8.33 \text{ ms}$$

Rearranging, we get  $\text{Conduction time} = 8.33 \text{ ms} - \text{delay time}$

And solving yields  $\text{Conduction time} = 8.33 \text{ ms} - 5 \text{ ms} = 3.33 \text{ ms}$

Therefore, the conduction time is 3.33 ms.

- (b) To find the conduction angle, use the relationship of Equation 4.8 to convert conduction time to conduction angle. For 60-Hz systems,

$$46.3 \mu\text{s} = 1 \text{ deg}, \text{ so the conversion factor is } \frac{1 \text{ deg}}{46.3 \mu\text{s}}$$

We can use this conversion factor to convert conduction time to conduction angle.

$$\text{Conduction angle} = 3.33 \text{ ms} \times \frac{1 \text{ deg}}{46.3 \mu\text{s}} \times \frac{1000 \mu\text{s}}{1 \text{ ms}} = 71.9 \text{ deg}$$

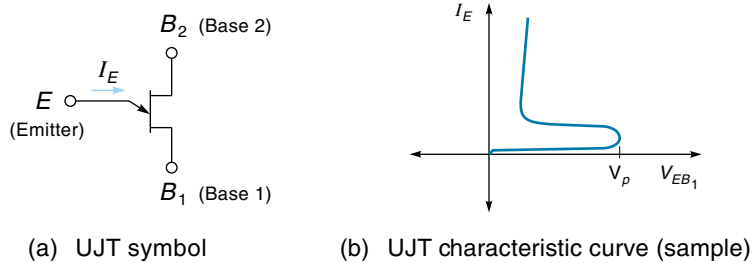
Therefore, the conduction angle is  $71.9^\circ$ .

## 4.6 TRIGGER DEVICES

The SCR and triac devices discussed in the previous sections both require **trigger circuits** to provide the gate turn-on signal. Ideally, this signal is a strong pulse delivered at the right time. Various semiconductor devices are particularly suitable for this job because of their bistable nature. This section gives a brief introduction to some of the most common triggering devices. For more information, consult a semiconductor devices text.

**Figure 4.37**

The unijunction transistor (UJT).



## Unijunction Transistors

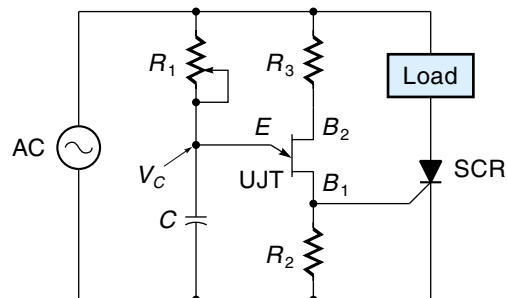
The **unijunction transistor** (UJT) symbol and characteristic curve are shown in Figure 4.37. Notice that it has three terminals: *base 1* ( $B_1$ ), *base 2* ( $B_2$ ), and the *emitter* ( $E$ ). The UJT operates as follows: For low emitter voltages ( $V_{EB1}$ ), there is no emitter current ( $I_E$ ). Once  $V_{EB1}$  reaches  $V_p$  (which is about two-thirds of the voltage between  $B_1$  and  $B_2$ ), the transistor fires, and current is allowed to surge through the emitter and  $B_1$ . UJTs are often used to trigger SCRs, as shown in the simplified circuit of Figure 4.38. For each positive half-cycle of the AC, the capacitor  $C$  charges up through  $R_1$ . When  $V_C$  gets high enough, the UJT fires, and the resulting current through  $B_1$  develops a large voltage across  $R_2$  and triggers the SCR.

## Programmable Unijunction Transistors

The **programmable unijunction transistor** (PUT) behaves like the UJT except that the firing voltage is adjustable via a resistor-divider network. The symbol and simplified trigger circuit are shown in Figure 4.39. Notice that it has three terminals: the **anode** (A), **cathode** (K), and **gate** (G). Like the SCR, the main current from A to K is held off until the anode voltage reaches  $V_p$  (peak voltage), at which point the PUT is triggered into conduction. The DC voltage on the gate determines what  $V_p$  will be. In the circuit shown in Figure 4.39(b),  $R_3$  and  $R_4$  form a resistor divider that establishes the gate voltage and hence the trigger point. The rest of the circuit is similar to other trigger circuits already discussed.

**Figure 4.38**

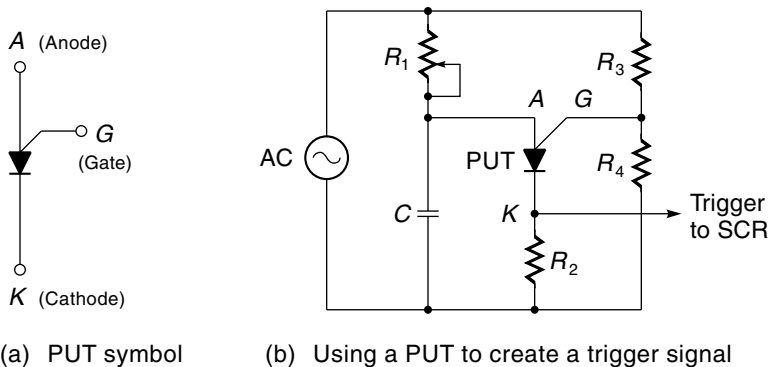
Using a UJT to trigger an SCR.





**Figure 4.39**

The programmable unijunction transistor (PUT).

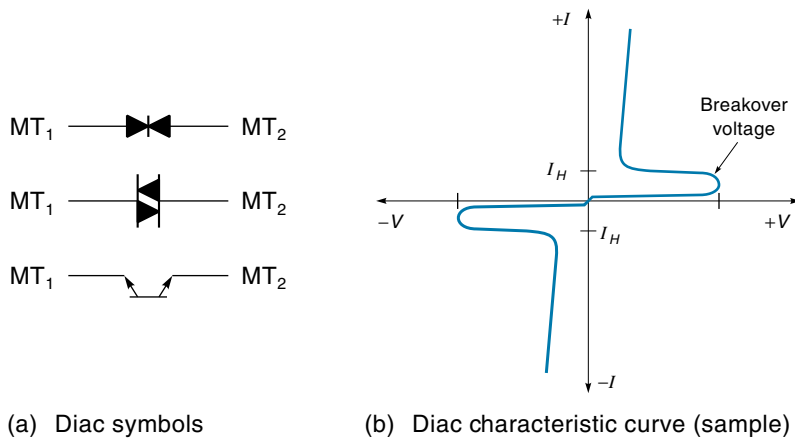


### Diac

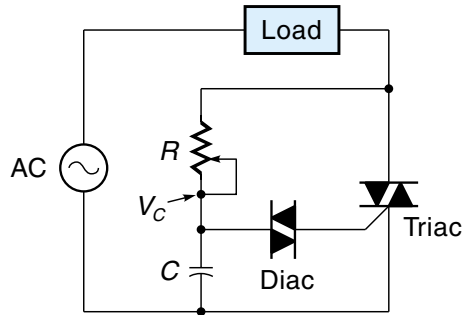
The **diac** symbol and characteristic curve are shown in Figure 4.40. It is a two-terminal bidirectional device, and it behaves essentially like a triac without a gate. As the curve indicates, if the voltage across the diac exceeds the breakover voltage (in either direction), it triggers into conduction mode and stays there until the the load current drops below the holding current. Diacs are frequently used to trigger triacs in full-wave AC applications, as shown in the simplified circuit of Figure 4.41. On the positive cycle of the AC, the capacitor will charge through  $R$ . When the voltage  $V_C$  rises to the positive breakover voltage of the diac, it (the diac) will trigger into conduction, which will then trigger the triac. Then, on the negative half-cycle of the AC, the capacitor will discharge through  $R$ , and when  $V_C$  drops to the negative breakover voltage of the diac, the diac will again trigger into conduction, which will trigger the triac. The *silicon bidirectional switch* (SBS) is similar to the diac except that it turns on more sharply and is more temperature stable. In addition, the SBS has a gate terminal that can be used or not; if used, it can alter the breakover voltage.

**Figure 4.40**

The diac.



**Figure 4.41**  
Using a diac to  
trigger a triac.



## SUMMARY

Switches are manually operated contacts that open and close. Many sizes and styles of switches are available, the most common being toggle, pushbutton, slide, DIP, rotary, thumbwheel, and membrane. Many of these styles are available with multiple contact configurations.

The traditional electromechanical relay (EMR) uses an electromagnet to open and close electrical contacts. Relay contacts are either normally open or normally closed, which refers to their unenergized state. As with switches, relays come in many sizes and contact configurations. Solid-state relays (SSRs) are solid-state switching circuits packaged so they can be used in place of EMRs in some applications.

The bipolar junction transistor (BJT), or simply *transistor*, is a three-terminal solid-state device that works like a valve to control the flow of current through a load (such as a motor). The transistor is a current amplifier, which means that the load current will be a multiple of the input base current. Another kind of transistor, called the field effect transistor (FET), performs the same job as the BJT, the main difference being that (for the FET) the load current is a function of the input gate voltage.

The silicon-controlled rectifier (SCR) is a three-terminal, bistable, solid-state device capable of switching very large DC currents. The SCR is easily switched on with a trigger pulse (to its gate terminal); however, the only way to turn it off is by reducing the load current to a very low value. Turning off the SCR requires additional circuitry for DC loads, but is not a problem in AC applications.

A triac is a three-terminal, bistable, solid-state device capable of switching AC current. The triac must be triggered into conduction each half-cycle with a trigger circuit. By delaying the trigger pulses, the amount of load power can be reduced, as in a lamp dimmer switch.

There are a number of smaller, bistable, solid-state devices that find uses in triggering SCRs and triacs. Examples of these devices are the unijunction transistor (UJT), the programmable unijunction transistor (PUT), and the diac.

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## GLOSSARY

**anode** One of three terminals of an SCR and PUT.

**base** One of three terminals of a transistor.

**biasing circuit** The part of a transistor circuit that generates the forward bias voltage.

**bipolar junction transistor (BJT)** Known as a *transistor*, a solid-state device that can be used as a switch or an amplifier.

**BJT** See **bipolar junction transistor**.

**cathode** One of three terminals of an SCR and PUT.

**class A operation** The event when a transistor is biased so the collector current is approximately half of its maximum value.

**class B operation** The event when a transistor is biased so that the collector current is just turning on.

**class C operation** The event when a transistor is biased below cutoff and used as an on-off switch.

**collector** One of the three terminals of a transistor.

**contactor** A heavy-duty relay that switches power directly to motors and machinery.

**Darlington** A three-terminal device (made of two transistors) that behaves like a transistor with a very high  $h_{FE}$  value.

**diac** A bistable, two-terminal, solid-state device that is used to trigger the triac.

**DIP switch** A set of SPST switches built into the shape of an IC (DIP stands for dual in-line package).

**double-pole/double-throw switch (DPDT)** A switch contact configuration.

**DPDT** See **double-pole/double-throw switch**.

**drain** One of three terminals in an FET.

**dv/dt effect** The event when the SCR turns on if the anode-cathode voltage rises too quickly.

**electromechanical relay (EMR)** A device that uses an electromagnet to open or close electrical contacts.

**emitter** One of three terminals of a transistor.

**EMR** See **electromechanical relay**.

**enhancement mode** Property of some MOSFETs where the gate voltage can always be positive (for N-channel).

**FET** *See* **field effect transistor**.

**field effect transistor (FET)** A three-terminal solid-state amplifying device that uses voltage as its input signal.

**forced commutation** The process of turning off an SCR by momentarily forcing the anode-cathode voltage to 0 V (or below).

**forward bias voltage** The DC offset base voltage required to start the transistor conducting.

**forward breakover voltage** The voltage across a thyristor that causes it to switch into its conduction state.

**forward conduction region** The operating range of a thyristor when it is conducting.

**forward current gain ( $h_{FE}$ )** The gain of a transistor; the collector current divided by the base current.

**gain-bandwidth product ( $f_T$ )** A constant for an amplifier that is the product of the open-loop gain and the frequency; can be read as the frequency when a transistor's gain has been reduced to 1.

**gate** One of three terminals in an SCR, PUT, FET, and triac.

**heat sink** A piece of metal, possibly with cooling fins, used to dissipate heat from a power device to the air.

**holding current ( $I_H$ )** The minimum anode current required to keep the thyristor in the conduction state.

**hybrid solid-state relay** A device that uses a reed relay to activate a triac.

**$I_{DSS}$**  The highest possible drain current for a particular JFET; occurs when the gate voltage is 0 V.

**JFET** *See* **junction FET**.

**junction FET** One type of an FET.

**limit switch** A switch used as a proximity sensor.

**membrane switch** Usually a keypad with a flexible membrane over the top.

**metal-oxide semiconductor FET (MOSFET)** One type of an FET.

**microswitch** A small push-button switch with a very short throw distance.

**minimum holding voltage** Minimum voltage needed to keep a relay activated.

**momentary-contact switch** A switch position that is spring-loaded.

**MOSFET** *See* **metal-oxide semiconductor FET**.

**MT<sub>1</sub>** One of three terminals of a triac.

**MT<sub>2</sub>** One of three terminals of a triac.

**normally open (NO), normally closed (NC)** The state of momentary switch or relay contacts when the device is not activated.

**NPN, PNP** The two basic types of transistors, the difference being the direction of current and voltage within the device.

**phase-control circuit** A circuit that can delay conduction for part of the AC cycle for the purpose of reducing average output voltage.

**power transistor** A transistor designed to carry a large current and dissipate a large amount of heat.

**programmable unijunction transistor (PUT)** A solid-state device that performs the same function as a UJT except that the trigger voltage is adjustable.

**pull-in current** Minimum current needed to activate a relay.

**pull-in voltage** Minimum voltage needed to activate a relay.

**push-button switch** A momentary switch activated by pushing.

**PUT** *See* **programmable unijunction transistor**.

**reed relay** A small relay with contacts sealed in a tube and activated by a magnetic field.

**rotary switch** A rotating knob that activates different switch contacts.

**SCR** *See* **silicon-controlled rectifier**.

**sealed current** Current required to keep a relay energized.

**shaded pole** A small metal ring around the end of the electromagnetic pole of an AC relay, for the purpose of keeping the relay from “buzzing” at 60 Hz.

**silicon-controlled rectifier (SCR)** A bistable, three-terminal, semiconductor device used to switch power to a load.

**single-pole/double-throw switch (SPDT)** A switch contact configuration.

**single-pole/single-throw switch (SPST)** A switch contact configuration.

**slide switch** Similar to a toggle switch except the handle slides back-and-forth.

**snubber** A circuit that prevents a fast voltage rise across an SCR, for the purpose of keeping it from false firing.

**solid-state relay (SSR)** A solid-state switching device used as a relay.

**source** One of three terminals in an FET.

**SPDT** *See* **single-pole/double-throw switch**.

**SPST** *See* **single-pole/single-throw switch**.

**SSR** *See* **solid-state relay**.

**switch wafer** Part of a rotary switch.

**three-position switch** Switch with a center position.

**thumbwheel switch** Switch that rotates a drum to select numeric data.

**thyristor** A class of four-layer semiconductor devices (such as PNP) that are inherently bistable; the SCR and triac are thyristors.

**toggle switch** A manually operated device that connects or disconnects power.

**transconductance ( $g_m$ )** The gain of an FET, which is the change in drain current divided by the change in gate voltage.

**triac** A bistable, three-terminal, solid-state device that switches power.

**trigger circuit** A circuit used to generate the turn-on pulse for an SCR or a triac.

**two-position switch** A toggle or slide switch with two positions.

**UJT** *See* **unijunction transistor**.

**unijunction transistor (UJT)** A bistable, three-terminal solid-state device that primarily triggers an SCR.

$V_{GS(off)}$  The gate voltage necessary to turn the drain current off (for a JFET).

**wiper** The center and/or moveable contact in a switch.

**zero-voltage switching** As applied to SSRs, the delay in switching until the AC voltage crosses the zero point.

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## EXERCISES

### Section 4.1

1. Draw the symbol of a DPDT toggle switch and explain how it operates.
2. Draw the symbol of the 1SCY191 switch (Table 4.1) and explain how it operates.
3. Draw the symbol of the 2SHX191 switch (Table 4.1) and explain how it operates.
4. For a rotary switch, draw a picture of the contact configuration for a one-pole/four-throw switch (terminal C makes contact with one of four terminals).

## Section 4.2

5. Draw the schematic symbol of a DPDT relay and explain how it works.
6. Draw the schematic symbol for the Y4-V52 relay (Table 4.2).
7. For relay Y4-V52 (Table 4.2), what are the voltage and current specifications for the coil and contacts?
8. For the relay Y6-V430 (Table 4.2), what are the voltage and current specifications for coil and contacts?
9. From the list of relays (Table 4.2), select a model to be used on an automobile to turn on a small spotlight (1.5 A).
10. List the advantages and disadvantages of an SSR versus an EMR.

## Section 4.3

11. A transistor has a current gain of 60 and a collector current of 5 A. Find the base current.
12. A transistor has a current gain of 40 and a base current of 25 mA. Find the collector current.
13. Calculate the exact and approximate emitter currents for the transistor described in Exercise 12.
14. A transistor with a gain of 35 has an input curve similar to that shown in Figure 4.16. The base voltage ( $V_{BE}$ ) is measured to be 0.8 V.
  - a. Find the base current.
  - b. Find the collector current.
15. For the transistor circuit shown in Figure 4.19, calculate  $I_C$  if the resistor  $R_1$  is changed to 1.3 k $\Omega$ . (Assume input curve Figure 4.16 applies.)
16. For the transistor circuit shown in Figure 4.19, calculate the  $I_C$  if the resistor  $R_2$  is changed to 120 k $\Omega$ . (Assume input curve Figure 4.16 applies.)
17. In regards to transistor amplifiers, explain the difference between class A, class B, and class C operations.
18. A 100-W power transistor is operating at a temperature of 60°C. What is the actual power capacity? (Assume a derate 0.57 W/°C above 25°C.)
19. Select a non-darlington transistor from the list in Figure 4.21 to meet at least the following specifications: Load current = 5 A, power dissipation = 60 W, and minimum gain = 15.
20. Draw a simple series circuit where an FET is used to control the power to a load and explain how it works. Include in your explanation the terms *gate voltage*, *drain-source current*, and  $g_m$ .
21. List the differences between a *JFET* and a *MOSFET*.
22. For the FET circuit shown in Figure 4.23, find the load current if the gate voltage is 6.5 V.
23. A circuit uses the MTW24N40E MOSFET (Figure 4.24) and has a gate voltage ( $V_{GS}$ ) of 5 V. Find the minimum and maximum drain current that could be expected in this circuit. (Assume  $g_m = 12$  mho.)

### Section 4.4

24. For an SCR, sketch the characteristic curve and use it to explain how the device operates.
25. List the three ways to turn an SCR off (for a DC circuit).
26. Explain why turning off the SCR is not a problem in AC circuits.
27. For a half-wave phase-control circuit, draw the waveforms for  $I_{AK}$  and  $V_{AK}$  if the SCR starts conducting at  $45^\circ$ .
28. For a full-wave phase-control circuit, draw the waveforms for  $I_{AK}$  and  $V_{AK}$  if the SCR starts conducting at  $135^\circ$ .
29. Select an SCR from the list in Figure 4.31 to meet the following specifications: Load current = 11 A and load voltage = 240 Vac. What is the trigger voltage required for your selected SCR?

### Section 4.5

30. Sketch the characteristic curve for the triac and use it to explain the operation of the device.
31. Draw the output waveform ( $I_m$ ) for a triac circuit that triggers at  $60^\circ$ .
32. Select a triac from the selection given in Figure 4.35 to meet the following specifications: Load current = 8 A and load voltage = 120 Vac. What is the trigger voltage required for your selected triac?
33. For a 60-Hz system, a triac has a conduction time of 2 ms. Find the delay time.
34. For a 60-Hz system, a triac has a conduction angle of  $45^\circ$ . Find the delay angle.
35. For a 60-Hz system, a triac has a delay time of 3.52 ms. Find the conduction time.
36. For a 60-Hz system, a triac has a conduction angle of  $85^\circ$ . Find the conduction time.

### Section 4.6

37. What is the function of  $R_1$  in the UJT circuit of Figure 4.38?
38. What is the purpose of  $R_3$  and  $R_4$  in the PUT circuit of Figure 4.39?
39. Explain how the diac triggers the triac in the circuit of Figure 4.41.



# CHAPTER 5

## Mechanical Systems

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### OBJECTIVES

After studying this chapter, you should be able to:

- Understand the properties of static, sliding, and viscous friction.
- Differentiate among the various types of springs and calculate the force that a spring exerts.
- Use the basic equations of linear and rotational motion to calculate the distance, velocity, and acceleration of an object acted on by a force.
- Convert the equivalent amounts of energy used in chemical, thermal, mechanical, and electrical systems and calculate energy-conversion efficiency.
- Understand the concept of heat conduction and perform simple heat-conduction calculations.
- Understand the properties of underdamped, overdamped, and critically damped mechanical systems.
- Calculate mechanical resonant frequencies.
- Understand the use of the various gear types and their terminology and perform gear train calculations.
- Know the characteristics of belts and roller chains used for power transfer.

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### INTRODUCTION

To understand the total electromechanical system, we need to understand some basic mechanical principles. In some ways, understanding the mechanics is easier than the electronics because at least you can see what the mechanical parts are doing. People have been watching things move all their lives—motors, gears, levers, springs, and so on, and have some feel for what is going on. In this chapter, we will first review the basic principles that dictate how mechanical systems respond to various forces and movements. Then the concepts of energy transfer, efficiency, and heat conduction will be introduced. We will not examine how strong a part is or how to design it—that is left to mechanical engineering textbooks.

An important part of any mechanical system is how power is transmitted from the power source to what is being driven. We will examine three ways this is done: gears, belts, and roller chains.

You may notice that many of the mechanical concepts have electrical analogs. For example, in a mechanical system, force causes movement; in an electrical system, voltage causes current. Friction is analogous to resistance, and mechanical resonance is similar to electrical resonance. The same set of principles can be applied in many areas.

A note about units: The United States is in the middle of the long slow process of switching from the English (or customary) units (inches, feet, pounds, etc.) to the SI (or metric) units (centimeters, meters, kilograms, etc.) Today, many machines in service, as well as many new machines, are based on the English system, but the trend is toward machines based on the metric system. This text uses primarily English units, but some examples are repeated using the SI units so that students can be exposed to both systems.

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## 5.1 BEHAVIOR OF MECHANICAL COMPONENTS

### Overview

An electronic controller outputs signals that are either analog or digital. When these signals are converted into mechanical motion, it becomes an electromechanical system, and we need to deal with a whole new set of conditions. Mechanical systems are subject to friction, flexing of parts, backlash, and effects of weight and inertia. The general effect of these factors is to slow the reaction time and/or make it difficult to move to a certain position with precision. Furthermore, mechanical parts tend to vibrate under certain conditions, which can cause damage to the parts as well as render the system useless.

### Friction

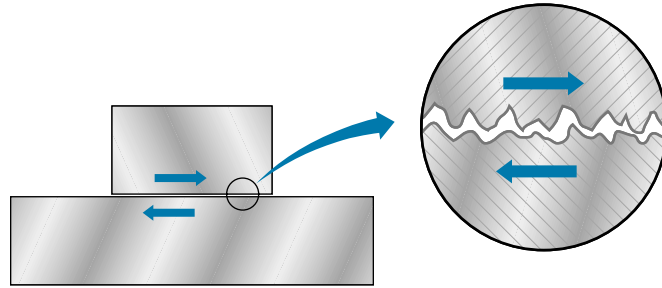
**Sliding friction**, the drag force that is always present when parts slide against each other, is caused by high spots on the sliding surfaces interfering with each other. Figure 5.1 illustrates this; the high spots tend to “catch” on each other and resist motion.

Friction force is proportional to the **normal force** ( $N$ ), which is the force pushing the parts together. The normal force could simply be the weight of the sliding part [Figure 5.2(a)], or the force could come from a bolt or spring, squeezing the sliding surfaces together, as in the partially tightened bolt in Figure 5.2(b). Often, parts need to move freely against each other, and this can present a dilemma. A snug fit is desirable for reducing rattles, but will have more friction than if the parts fit loosely.

Friction is greatest at startup. It takes more force to start a part moving than it does to keep it moving, sometimes twice as much. Startup friction is known as **static friction**;

**Figure 5.1**

Friction is caused by uneven surfaces “catching.”



it is particularly troublesome for control systems because, once enough force is applied to overcome the static friction, the resistance immediately drops and the part tends to overshoot its destination. We will discuss how to deal with this problem in Chapter 11.

As you might imagine, the actual value of force necessary to overcome friction depends on the type of materials involved as well as on the normal force; soft rubber has more friction than steel. Materials can be tested for friction and assigned a value called the **coefficient of friction** ( $\mu$ ). Table 5.1 gives the coefficient of friction for some common materials, but note that these values can vary widely from test to test and should be considered approximate. Using the coefficient of friction and the normal force, we can calculate the force to overcome friction from Equation 5.1.

$$F = \mu N \quad (5.1)$$

where

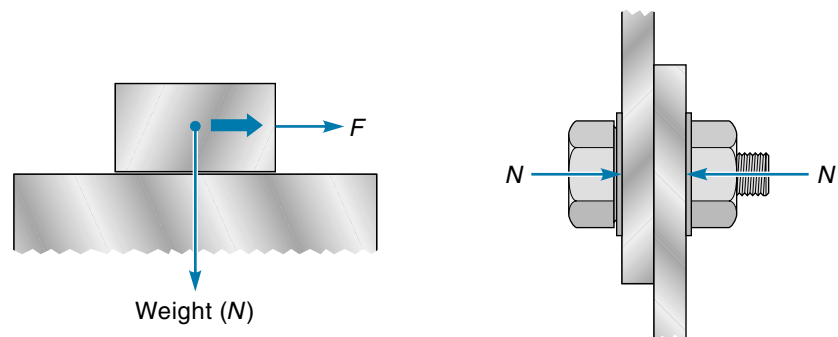
$F$  = force to overcome friction

$\mu$  = coefficient of friction (dependent on material)

$N$  = normal force—that is, force pushing materials together

**Figure 5.2**

Friction is proportional to normal force ( $N$ ) caused by (a) a weight and (b) bolt tightening.

(a) Friction ( $F$ ) proportional to weight

(b) Friction from tightening bolt

TABLE 5.1

## Coefficient of Friction for Various Materials\*

Material	Static	Sliding
Steel on steel	0.78	0.42
Aluminum on steel	0.61	0.47
Glass on glass	0.94	0.40
Oak on oak	0.62	0.48
Rubber on pavement	0.9	0.85
Diamond on diamond	0.2	
Teflon on Teflon	0.04	

\*from Marks *Standard Handbook for Mechanical Engineers* 1967 and others

## EXAMPLE 5.1

A steel block weighing 1.5 lb is resting on a steel surface.

- How much force is necessary to get it started sliding?
- How much force is necessary to keep it sliding once it is moving?

## SOLUTION

If the block weighs 1.5 lb, then the force between the block and the steel surface is 1.5 lb, which means that the normal force is 1.5 lb. To solve this problem, we will employ Equation 5.1 twice, the first time using the coefficient of friction for static friction, the second time using the value for sliding friction.

To move the block from rest, use the *static-friction* value for  $\mu$  (for steel).

$$F = \mu N = 0.78 \times 1.5 \text{ lb} = 1.17 \text{ lb}$$

To keep the block sliding once it is in motion, use the *sliding-friction* value for  $\mu$ .

$$F = \mu N = 0.45 \times 1.5 \text{ lb} = 0.68 \text{ lb}$$

Therefore, we can say that 1.17 lb of force is necessary to get the block moving from a standstill, but only 0.68 lb of force is necessary to keep it sliding once it is moving.

Lubrication between the parts can greatly reduce the friction and change its character. Ideally, the lubricant keeps the parts from actually touching each other, and the movement is supported by layers of lubricant slipping over each other. The drag on the part from the lubricant is known as **viscous friction**, and its force is directly propor-

tional to the relative velocity between the moving parts. (This is different from sliding friction where the drag force is relatively constant.) A good lubricant must be sticky enough to adhere to the material, but not so sticky as to impede motion. A potential problem is that contaminants in the area such as dirt and sand will tend to stick in the lubricant, which will increase the friction and possibly damage the parts. Some applications use a “dry” lubricant, which is available in a spray can and is applied as a thin, nonstick coating on a part. Some mechanical joints are designed to run “dry” without any lubricant; therefore, check the manual before you lubricate with oil every moving joint you see.

Rolling friction produces considerably less drag than sliding friction. For example, using a wagon to carry something requires less “pulling force” than dragging it. Ball bearings and roller bearings (Figure 5.3) make use of this property to achieve very low friction values.

That part of energy expended overcoming friction is converted to heat. Any system that transports mechanical energy, such as a ball bearing or a gearbox, has a certain efficiency. The efficiency tells what percentage of the mechanical power going in actually comes out, with the rest being lost to friction. A ball bearing can be 99% efficient, a bicycle chain is about 96% efficient, but a worm gear may be only 60% efficient. (Energy and efficiency are discussed later in this chapter.)

## Springs

Common components in mechanical systems, springs absorb shock (for example, automobile chassis springs), store energy in spring-wound motors, and provide a constant pressure (for example, in a clothespin).

**Figure 5.3**

A ball bearing and a roller bearing. (Roller bearing courtesy of The Timken Co.)



(a) Ball bearing



(b) Roller bearing

The force that a spring produces can be predicted using **Hooke's law**, which the following states mathematically:

$$F = kx \quad (5.2)$$

where

$F$  = push or pull force of the spring

$k$  = spring constant, different for each size of spring

$x$  = distance spring is extended from its "rest length"

Equation 5.2 tells us that the more a spring is extended or compressed, the more force it produces. Because force increases linearly, to get twice the force, you must extend it twice as far. This relationship has a limit, of course. At some point, the spring will deform and never return to its original shape. The use of Hooke's law is illustrated in Example 5.2.

### EXAMPLE 5.2

A coil spring with a spring constant of 10 lb/in. has a rest length of 2.0 in. In a machine, it provides a constant pressure on a belt-tensioning pulley (Figure 5.4). The spring length in the machine is 2.5 in. How much force is the spring providing?

### SOLUTION

First, we calculate the distance the spring is extended by finding the difference between the extended length and the rest length:

$$x = 2.5 \text{ in.} - 2.0 \text{ in.} = 0.5 \text{ in.}$$

Now use the spring constant ( $k$ ) of 10 lb/in. in Equation 5.2 to calculate the force the spring exerts on the pulley:

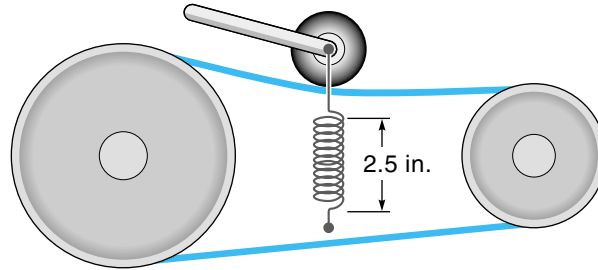
$$\begin{aligned} F &= kx \\ &= \frac{10 \text{ lb}}{\text{in.}} \times 0.5 \text{ in.} = 5 \text{ lb} \end{aligned}$$

The spring is exerting a pressure of 5 lb on the tensioning pulley.

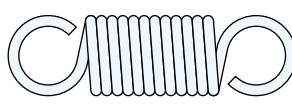
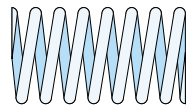
Many different types of springs are in use. Perhaps the most common is the *coil spring*; coil springs can be either extension or compression types [Figure 5.5(a)]. *Extension springs* may have completely closed coils at rest. They are used to provide a pull as described in Example 5.2. *Compression springs* usually have flat ends and are

**Figure 5.4**

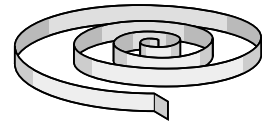
A spring used to provide tension on a belt-tensioning pulley (Example 5.2).

**Figure 5.5**

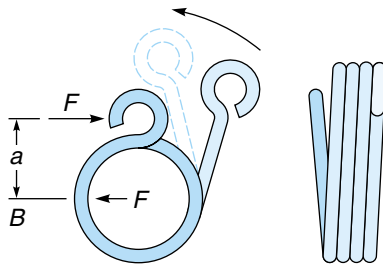
Types of springs.



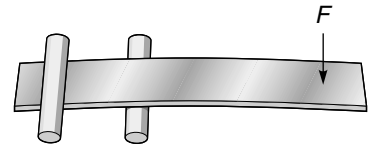
(a) Compression spring and extension spring



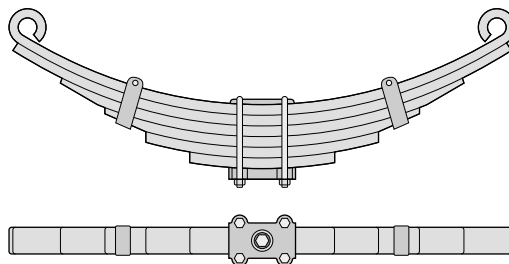
(b) Motor spring



(c) Torsion spring



(d) Flat spring



(e) Leaf spring

intended to be compressed. Examples of these are valve springs in an engine and the small springs under each key in a keyboard.

Figure 5.5(b) illustrates a *motor spring*, a flat strip rolled in a coil with one end fixed. The energy-storage capacity of a spring is not as high as a battery for equal weights; however, a spring will not discharge “on the shelf,” and its output is direct mechanical motion. *Torsion springs* are used to provide torque or twist. Figure 5.5(c) illustrates one example. A typical clothespin uses a torsion spring. A *flat spring*, a thin flat piece of metal [Figure 5.5(d)], is a simple device used extensively in applications where a constant pressure is needed and the deflection distance is low. A *leaf spring* [Figure 5.5(e)] is made from several large flat springs. The leaf spring has the combined load strength of the individual flat springs.

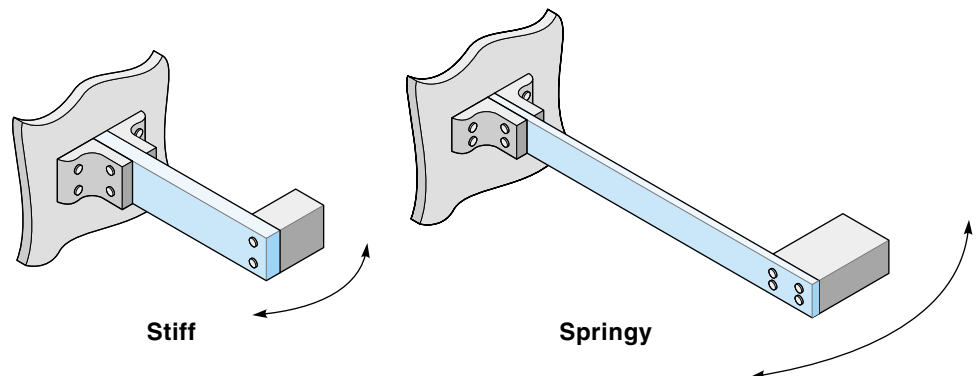
Finally, it is important to realize that *every* mechanical part is a spring, whether it was intended to be or not. Any piece of material (particularly metal) will deflect under pressure and then spring back when the pressure is removed. The actual spring constant of a part depends on its composition, its shape, and how the force is applied. In general, a long thin part is much more springy than a shorter thicker part (Figure 5.6). As we will see later in this chapter, the springiness of system components can adversely affect the performance of a machine.

## Mass and Inertia

Every mechanical part has a **mass**, which is a measure of the amount of matter present. The primary unit of mass is the kilogram (kg), (there is no common unit of mass in customary units). Related to mass is the familiar concept of weight. **Weight** is the downward force that an object exerts because of gravitational attraction. Weight is measured in units of force, such as the pound (lb). In the SI system, force is measured in **Newtons (N)**, where 1 N is equivalent to  $1 \text{ kg} \cdot \text{m/s}^2$ . The Newton is equal to a force of 0.224 lb.

**Figure 5.6**

A long, thin part is more springy than a short, thick part.





On Earth, mass and weight are directly proportional to each other and are often used interchangeably by the general public. For example, in Europe, people buy oranges by the kilogram, whereas oranges are sold by the pound in the United States. However, from an engineering standpoint, mass and weight (force) are different concepts and must be treated as such. Newton's second law of motion (Equation 5.3) specifies the relationship between mass and force, telling us that when a force is applied to a mass, the mass will accelerate:

$$F = ma \quad (5.3)$$

where

$F$  = force applied

$m$  = mass of the object being moved

$a$  = acceleration of the object

The relationship between weight and mass can be expressed as a special case of Equation 5.3, given as

$$w = mg \quad (5.3a)$$

where

$w$  = weight (downward force due to gravity)

$m$  = mass

$g$  = acceleration of gravity (32 ft/s<sup>2</sup> or 9.8 m/s<sup>2</sup>)

In Equation 5.3a, weight has been substituted for force, and the gravitational constant ( $g$ ) has been substituted for acceleration. As you will see, Equation 5.3a will be used when solving numerical problems.

Momentum is a property of a moving object. It takes more effort to stop a rolling car than a child's wagon—we say that the moving car has more momentum. **Momentum** is defined as mass times velocity and is a measure of how much energy it takes to get a part moving. This same amount of energy must be dissipated when the part is stopped.

$$\text{Momentum} = mv \quad (5.4)$$

where

$m$  = mass of the moving object

$v$  = velocity of the object

**Inertia**, a basic characteristic of mass, is the tendency of an object to remain at rest or in motion unless acted on by an external force. An example would be throwing a ball. Once the ball leaves your hand, it continues to move through the air because of inertia.

**Figure 5.7**

A demonstration of force and inertia.

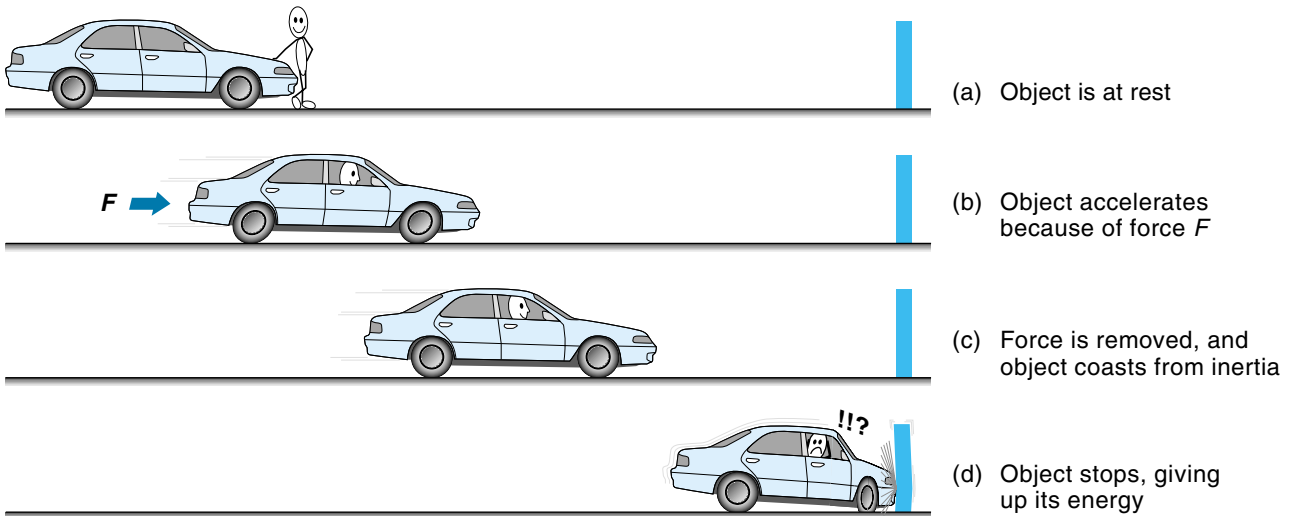


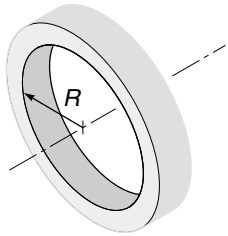
Figure 5.7 uses a car to illustrate the basic concepts of linear motion (moving in a straight line).

Inertia is also demonstrated when an object rotates about an axis. Consider a bicycle wheel and a car wheel. Both are about the same diameter, but the heavier car wheel has much more inertia and therefore takes more energy to get spinning. However, the inertia of a rotating part is a function of *shape and mass*, not just mass alone, and is given a special name: **moment of inertia** ( $I$ ). Figure 5.8 shows a few shapes. In general, parts that have the bulk of their mass at a distance from the axis have more rotational inertia than parts where the bulk of the mass is close to the axis. Equations are available to calculate the moment of inertia for different shapes. Notice that the flywheel in Figure 5.8(a) has far more rotational inertia than the solid wheel of Figure 5.8(b), even though both wheels are the same mass. This is because most of the mass in the flywheel is concentrated as far from the axis as possible. Figure 5.8(c) shows the rotational inertia in a long beam. A part does not have to look like a wheel to have rotational inertia; the only requirement is that it rotates about an axis. In Figure 5.8(c), most of the mass is farther from the axis, so the moment of inertia is higher than the situation in Figure 5.8(d) where the axis is through the middle.

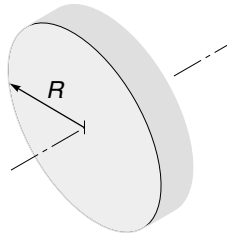
In some situations, high rotational inertia is desirable, in which case a flywheel is used. For example, in a car engine, the flywheel smooths out the power impulses from the individual pistons. However, in most cases, inertia creates problems for electro-mechanical control systems because a high-inertia part takes more force to get moving and more force to stop. This results in slower response times and positional inaccura-

**Figure 5.8**

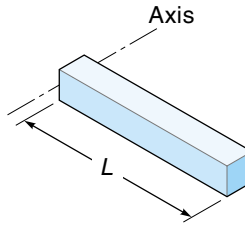
Moments of inertia for four shapes.



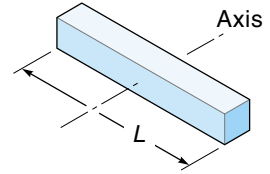
(a) Flywheel  $I = MR^2$



(b) Solid wheel  $I = \frac{1}{2} MR^2$



(c) Axis at end  $I = \frac{1}{3} ML^2$



(d) Axis in middle  $I = \frac{1}{12} ML^2$

cies. The question is, Why not design parts to have less inertia? The problem is that inertia is a function of weight, and weight is a function of strength. Obviously, a part must be strong enough to carry the load, so a certain amount of inertia is unavoidable.

### Basic Equations of Motion for Linear Systems

A set of basic equations allows you to calculate the position, velocity, and the time it takes to get someplace, for an object being acted on by a force. Perhaps you have seen and used these before in a physics class, but it is worthwhile to repeat here because we will be using them from time to time in the text to explain mechanical behavior.

We will start by repeating Equation 5.3:

$$F = ma$$

This equation is used to calculate the force required to accelerate an object. The velocity of an object that is being uniformly accelerated can be computed from Equation 5.5. Remember that a uniform **acceleration** means that the velocity is increasing at a constant rate:

$$v = at \quad (5.5)$$

where

$v$  = velocity

$a$  = acceleration

$t$  = time it takes to go from rest to velocity  $v$

The distance that an object with a constant velocity has traveled in a certain time can be computed as follows:

$$d = vt \quad (5.6)$$

where

$d$  = distance the object has moved

$v$  = velocity of the object

$t$  = time the object has been moving

The distance that an object under uniform acceleration has traveled in a certain time is computed as follows:

$$d = \frac{1}{2}at^2 \quad (5.7)$$

where

$d$  = distance the object has moved

$a$  = acceleration

$t$  = time since acceleration started

Finally, we can use Equation 5.8 to calculate the velocity if the distance over which an object has been accelerating is known:

$$v = \sqrt{2ad} \quad (5.8)$$

where

$v$  = velocity

$a$  = acceleration

$d$  = distance the object has moved under acceleration

### EXAMPLE 5.3

A pneumatic cylinder pushes defective parts off a conveyer belt (Figure 5.9). A part could weigh up to 10 lb, and it must be pushed a distance of 1 ft in 1 s or less. What constant force of the cylinder would be required to do the job? (Pneumatics is covered in Chapter 10.)

### SOLUTION

We are given the distance a part has to be moved and the time to do it. None of the preceding equations relate distance, time, and force directly, but Equation 5.7 relates time, distance, and acceleration. Once we know the acceleration required, we can calculate the force required, using Equation 5.3. So, starting with Equation 5.7,

$$d = \frac{1}{2}at^2$$

We first solve for  $a$ :

$$a = \frac{2d}{t^2} = \frac{2 \times 1 \text{ ft}}{1 \text{ s}^2} = 2 \text{ ft/s}^2$$

Knowing the acceleration, we now calculate the force required to give that acceleration, using Equation 5.3:

$$F = ma$$

But first we must calculate mass from weight using Equation 5.3a.

$$m = \frac{w}{g} = \frac{10 \text{ lb}}{32 \text{ ft/s}^2} = 0.31 \frac{\text{lb} \cdot \text{s}^2}{\text{ft}}$$

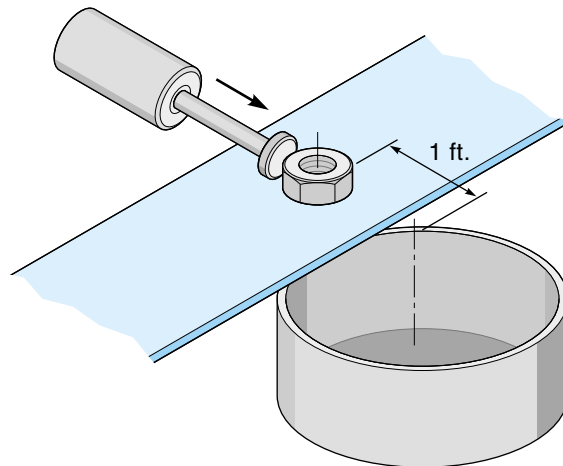
Finally, we calculate the required force of the cylinder from Equation 5.3:

$$F = ma = \frac{0.31 \text{ lb} \cdot \text{s}^2}{\text{ft}} \times \frac{2 \text{ ft}}{\text{s}^2} = 0.62 \text{ lb}$$

The cylinder must push with a force of at least 0.62 lb.

**Figure 5.9**

The pneumatic cylinder must push a defective part off the conveyor belt in 1 s or less (Example 5.3).



**EXAMPLE 5.3 (Repeated with SI Units)\***

A pneumatic cylinder pushes defective parts off a conveyer belt (Figure 5.9). A part could have a mass of up to 5 kg, and it must be pushed a distance of 25 cm in 1 s or less. What is the force of the cylinder?

**SOLUTION**

Calculate the required acceleration of the part, using Equation 5.7:

$$d = \frac{1}{2}at^2$$

Next, solve for  $a$ :

$$a = \frac{2d}{t^2} = \frac{2 \times 25 \text{ cm}}{1 \text{ s}^2} = \frac{50 \text{ cm}}{\text{s}^2} = \frac{0.5 \text{ m}}{\text{s}^2}$$

Knowing the acceleration, we now calculate the force required to give that acceleration, using Equation 5.3:

$$\begin{aligned} F &= ma \\ &= 5 \text{ kg} \times \frac{0.5 \text{ m}}{\text{s}^2} = 2.5 \text{ N} \end{aligned}$$

The cylinder must push with a force of at least 2.5 N.

**EXAMPLE 5.4**

An electric solenoid is used to drive the print hammer in a high-speed printer. The hammer presses the print type into the inked ribbon, which prints on the paper (Figure 5.10). The hammer weighs 0.1 lb. It must strike the type with a velocity of 60 in./s and moves through a distance of 0.5 in. How much force must the solenoid provide to the hammer?

**SOLUTION**

In this problem, we are given distance and velocity and need to find the force pushing the hammer. As in Example 5.3, this is a two-step problem: We first need to find the acceleration required and then the force to create that acceleration. We will modify Equation 5.8 to compute the acceleration, but we first need to convert weight to mass:

\*In this chapter the values used in the SI examples are similar to those used with the English units, but not exactly the same.

$$m = \frac{w}{g} = \frac{0.1 \text{ lb}}{32 \text{ ft/s}^2} = 0.00313 \frac{\text{lb} \cdot \text{s}^2}{\text{ft}}$$

Rearranging Equation 5.8 to solve for acceleration, we get

$$v = \sqrt{2ad}$$

$$a = \frac{v^2}{2d} = \frac{(60 \text{ in./s}^2)^2}{2 \times 0.5 \text{ in.}} = \frac{3600 \text{ in.}^2}{1 \text{ in.} \cdot \text{s}^2} = 3600 \frac{\text{in.}}{\text{s}^2}$$

Now, from Equation 5.3, we can calculate the force to push the hammer:

$$F = ma = \frac{0.00313 \text{ lb} \cdot \text{s}^2}{\text{ft}} \times \frac{3600 \text{ in.}}{\text{s}^2} \times \frac{1 \text{ ft}}{12 \text{ in.}} = 0.939 \text{ lb}$$

The solenoid would have to provide a force on the hammer of at least 0.939 lb.

#### EXAMPLE 5.4 (Repeated with SI Units)

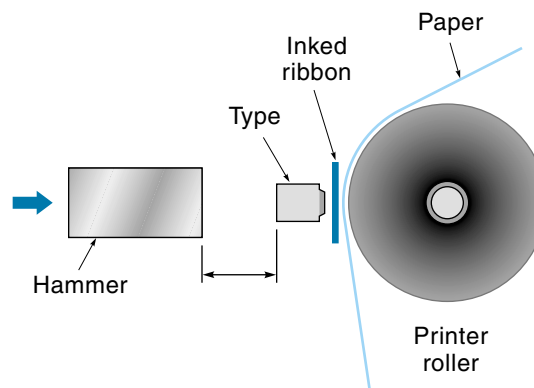
An electric solenoid is used to drive the print hammer in a high-speed printer. The hammer presses the print type into the inked ribbon, which prints on the paper (Figure 5.10). The mass of the hammer is 0.05 kg (50 g). It must strike the type with a velocity of 1.5 m/s and moves through a distance of 1 cm. How much force must the solenoid provide to the hammer?

#### SOLUTION

In this problem, we are given distance and velocity and need to find the force pushing the hammer. As in Example 5.3, this is a two-step problem: We first

**Figure 5.10**

The print hammer must hit the type at a specified velocity (Example 5.4).



need to find the acceleration required and then the force to create that acceleration. To solve for acceleration, rearrange Equation 5.8:

$$v = \sqrt{2ad}$$

$$a = \frac{v^2}{2d} = \frac{(1.5 \text{ m/s})^2}{2 \times 1 \text{ cm}} = \frac{2.25 \text{ m}^2}{2 \text{ cm} \cdot \text{s}^2} \times \frac{100 \text{ cm}}{\text{m}} = \frac{112.5 \text{ m}}{\text{s}^2}$$

Now, from Equation 5.3, we can calculate the required hammer force:

$$F = ma = 0.05 \text{ kg} \times 112.5 \frac{\text{m}}{\text{s}^2} = 5.6 \text{ N}$$

The solenoid would have to provide a force on the hammer of at least 5.6 Newtons.

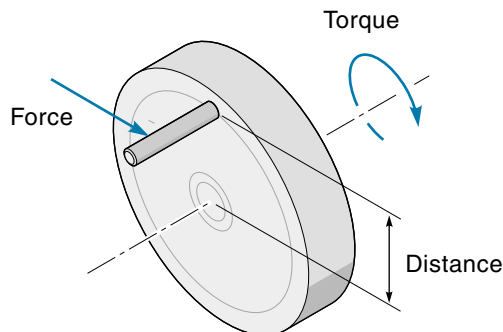
## Basic Equations of Motion for Rotational Systems

Rotational motion refers, of course, to things that go around. Wheels, gears, axles, and motors all follow the laws of rotational motion. The basic equations for rotational systems are almost the same as those for linear motion. Every quantity used in linear motion has an analog in rotational motion. Instead of force, we have torque. Instead of acceleration and velocity, we have angular acceleration and angular velocity; instead of mass, we have moment of inertia ( $I$ ). Moment of inertia, as discussed earlier, is dependent on mass, shape, and location of the axis. Equations for calculating  $I$  for a few standard shapes are given in Figure 5.8. Equations for  $I$  for other shapes are readily available in handbooks.

Torque is the kind of force motors produce. **Torque** is a twisting force acting at a certain radial distance from the center of rotation (Figure 5.11). The equation for torque follows:

**Figure 5.11**

Torque is caused by a force ( $F$ ) acting at a distance from the axis.





$$T = Fd \quad (5.9)$$

where

$T$  = torque

$F$  = force

$d$  = distance of force from the axis

The basic law of rotational motion is given in Equation 5.10 and has the same form as  $F = ma$ . It tells us that if a constant torque is applied to a wheel, that wheel will accelerate its rotation:

$$T = I\alpha \quad (5.10)$$

where

$T$  = torque applied to a rotating object

$I$  = moment of inertia for the object

$\alpha$  = angular acceleration of the object

The angular position of an object rotating at a constant angular velocity can be calculated from Equation 5.11:

$$\theta = \omega t \quad (5.11)$$

where

$\theta$  = angular position in radians\*

$\omega$  = angular velocity

$t$  = time the object has been rotating

The angular velocity of an object under constant angular acceleration can be calculated from Equation 5.12:

$$\omega = \alpha t \quad (5.12)$$

where

$\omega$  = angular velocity

$\alpha$  = angular acceleration

$t$  = time the object has been accelerating

---

\*A radian is a sort of natural unit of angle. Numerically speaking, there are  $2\pi$  radians in  $360^\circ$ , or  $\pi$  radians in  $180^\circ$ , or 1 radian =  $57.3^\circ$ . Angles must be converted to radians when these equations are applied (otherwise, the equations won't work).

The angular position of an object under constant acceleration can be calculated from Equation 5.13:

$$\theta = \frac{1}{2}\alpha t^2 \quad (5.13)$$

The angular velocity of an object that has moved to a certain position with a certain angular acceleration can be calculated from Equation 5.14:

$$\omega = \sqrt{2\alpha\theta} \quad (5.14)$$

### EXAMPLE 5.5

A motor is used to open and close a damper in an air duct (Figure 5.12.) The motor can produce a torque of 30 in. · lb, and the moment of inertia ( $I$ ) of the damper is 0.8 lb · s<sup>2</sup> · in. How long will it take the damper to open, and what is the maximum angular velocity of the damper? In this simple system, the motor comes on full torque and stays on until the damper hits a limit switch.

### SOLUTION

Because we know how much torque the motor can supply and the  $I$  of the load, we first calculate the angular acceleration of the damper from Equation 5.10:

$$T = I\alpha$$

Solving for  $\alpha$ , we get

$$\alpha = \frac{T}{I} = \frac{30 \text{ in.} \cdot \text{lb}}{0.8 \text{ lb} \cdot \text{s}^2 \cdot \text{in.}} = \frac{37.5 \text{ rad}}{\text{s}^2}$$

We now have the acceleration, but we are looking for the time the damper takes to open (that is, time for the damper to rotate 90°). Equation 5.13 relates time, position, and acceleration, but position must be in radians, so we first convert 90° to radians. Recalling that there are  $\pi$  radians in 180°, we find that

$$90^\circ = 90^\circ \times \frac{\pi \text{ rad}}{180^\circ} = 1.57 \text{ rad}$$

Now we use Equation 5.13,  $\theta = (1/2)\alpha t^2$ , and solve for  $t$ :

$$t = \sqrt{\frac{2\theta}{\alpha}} = \sqrt{\frac{2 \times 1.57 \text{ rad} \cdot \text{s}^2}{37.5 \text{ rad}}} = 0.29 \text{ s}$$

The calculations show that the damper will open in about one-third of a second. We were also asked to find the maximum angular velocity of the damper. Because the motor provides a constant torque, the damper is accelerating all

the time it is opening, and the maximum velocity will occur at 0.29 s. We already calculated the acceleration and time, so we will use Equation 5.12 because it relates velocity, acceleration, and time:

$$\omega = \alpha t = \frac{37.5 \text{ rad}}{\text{s}^2} \times 0.29 \text{ s} = 10.87 \frac{\text{rad}}{\text{s}}$$

Converting to degrees/second gives us

$$\frac{10.87 \text{ rad}}{\text{s}} \times \frac{180^\circ}{\pi \text{ rad}} = \frac{623^\circ}{\text{s}}$$

Therefore, the fastest angular velocity of the damper is 623°/s.

### EXAMPLE 5.5 (Repeated with SI Units)

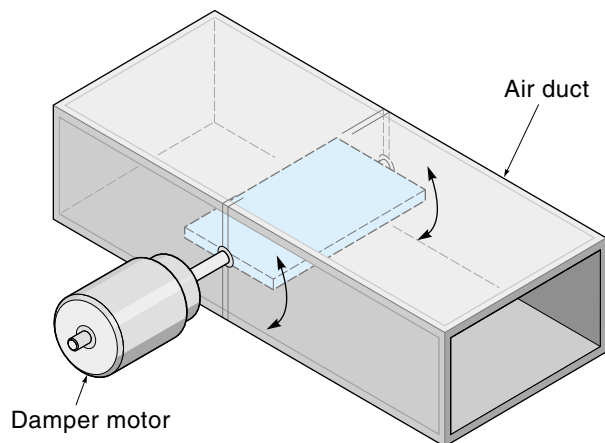
A motor is used to open and close a damper in an air duct (Figure 5.12). The motor can produce a torque of 3 N · m, and the moment of inertia ( $I$ ) of the damper is 0.08 kg · m<sup>2</sup>. How long will it take the damper to open, and what is the maximum angular velocity of the damper? In this simple system, the motor comes on full torque and stays on until the damper hits a limit switch.

### SOLUTION

First, calculate the angular acceleration of the damper from Equation 5.10:

$$T = I\alpha$$

**Figure 5.12**  
A damper in  
an air duct  
(Example 5.5).



Solving for  $\alpha$ , we get

$$\alpha = \frac{T}{I} = \frac{3 \text{ N} \cdot \text{m}}{0.08 \text{ kg} \cdot \text{m}^2} = \frac{37.5 \text{ rad}}{\text{s}^2}$$

Now, use Equation 5.13 to calculate the time it takes the damper to open (rotate  $90^\circ$ ). However, we must first convert  $90^\circ$  to radians. Recalling that there are  $\pi$  radians in  $180^\circ$ , we find that

$$90^\circ = 90^\circ \times \frac{\pi \text{ rad}}{180^\circ} = 1.57 \text{ rad}$$

Now use Equation 5.13,  $\theta = (1/2)\alpha t^2$ , and solve for  $t$ :

$$t = \sqrt{\frac{2\theta}{\alpha}} = \sqrt{\frac{2 \times 1.57 \text{ rad} \cdot \text{s}^2}{37.5 \text{ rad}}} = 0.29 \text{ s}$$

The calculations show that the damper will open in about one-third of a second. Use Equation 5.12 to calculate the maximum rotational velocity, which will occur at 0.29 s:

$$\omega = \alpha t = \frac{37.5 \text{ rad}}{\text{s}^2} \times 0.29 \text{ s} = 10.87 \frac{\text{rad}}{\text{s}}$$

Converting to degrees/second gives us

$$\frac{10.87 \text{ rad}}{\text{s}} \times \frac{180}{\pi \text{ rad}} = \frac{623^\circ}{\text{s}}$$

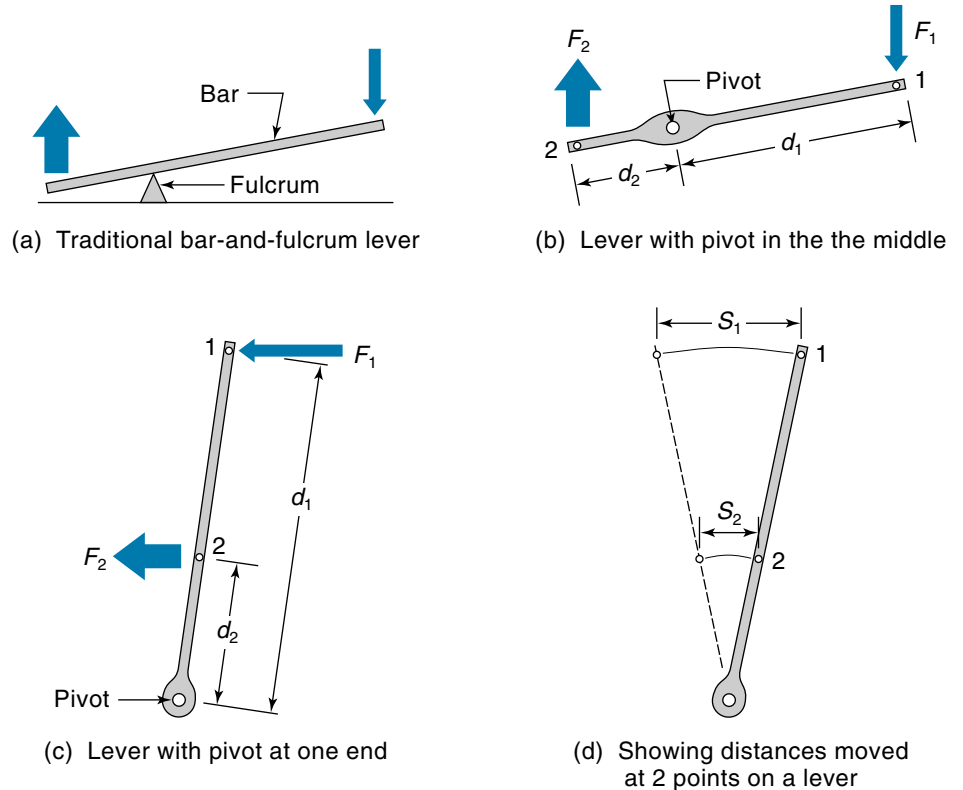
Therefore, the fastest angular velocity of the damper is  $623^\circ/\text{s}$ .

## Levers

One of the simplest ways to amplify force is with a lever. A traditional lever is a bar supported by a fulcrum (pivot point) [see Figure 5.13(a)]. Archimedes is supposed to have boasted that given a lever long enough and a place to stand, he could lift the world. In mechanical systems, a **lever** is a rigid bar pivoted somewhere in the middle or at one end and acted on by two opposing forces [see Figure 5.13(b and c)]. In either case, a force  $F_1$  applied to the lever at point 1 is converted to a larger force ( $F_2$ ) at point 2 (where point 2 is closer to the pivot than point 1). The actual force at  $F_2$  depends on the ratio of distances  $d_1$  and  $d_2$ . The principle works like this: in Figure 5.13(b), the force  $F_1$  is actually providing a torque CW (clockwise) around the pivot ( $T_1 = F_1 \times d_1$ ). If this were the only torque on the lever, it would start to spin (accelerate) in a CW direction. Thus, if the lever is *not* accelerating in a CW direction (which it isn't), there must be

Figure 5.13

Levers



an equal and opposite torque canceling out  $T_1$ . The opposing torque ( $T_2$ ) must be coming from the load resisting  $F_2$ .

If the opposing torques balance each other,  $T_1 = T_2$ . And therefore,  $F_1 \times d_1 = F_2 \times d_2$ . Solving for  $F_2$  yields

$$F_2 = \frac{F_1 \times d_1}{d_2} \quad (5.15^*)$$

where  $F_2$  = force available at point 2  
 $F_1$  = force applied at point 1  
 $d_1$  = distance from pivot to  $F_1$   
 $d_2$  = distance from pivot to  $F_2$

When levers are used in machines, they usually oscillate back and forth. The linear distance that a point on the lever moves is called its *stroke*. As you can clearly see from Figure 5.13(d) the stroke of point 1 ( $s_1$ ) is larger than the stroke of point 2 ( $s_2$ ).

\*This formula assumes that  $F_1$  and  $F_2$  are parallel to each other.

Actually, the trajectories of points 1 and 2 are arcs, but if the movements are small enough we can approximate them as straight lines. Using simple geometry, it can be shown that

$$s_2 = \frac{s_1 \times d_2}{d_1} \quad (5.16)$$

where  $s_2$  = stroke (distance of travel) of point 2  
 $s_1$  = stroke (distance of travel) of point 1  
 $d_1$  = distance from pivot to  $F_1$   
 $d_2$  = distance from pivot to  $F_2$

### EXAMPLE 5.6

The clutch pedal of a car is shown in Figure 5.14. To depress the clutch pedal requires 5 lb of pedal pressure and a movement of 2 inches. Find the force and stroke on the clutch linkage shaft.

### SOLUTION:

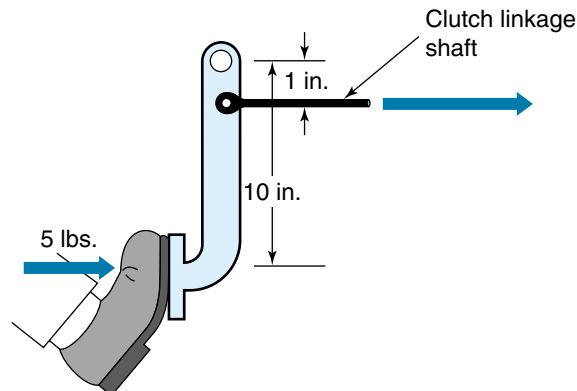
The clutch pedal is attached to one end of a lever, and the other end of the lever is pivoted up under the dashboard. The driver's foot applies force  $F_1$ , and a larger force  $F_2$  is applied to the clutch linkage (but the stroke will be shorter). The distance from the pedal to the pivot ( $d_1$ ) is 10 in., and the distance from the linkage take-off to the pivot ( $d_2$ ) is 1 in.

Use Equation 5.15 to find the force applied to the linkage.

$$F_2 = \frac{F_1 \times d_1}{d_2} = \frac{5 \text{ lb} \times 10 \text{ in.}}{1 \text{ in.}} = 50 \text{ lb}$$

Use Equation 5.16 to find the stroke of the linkage.

**Figure 5.14**  
Lever action used  
in a clutch pedal  
(Example 5.6).



$$s_2 = \frac{s_1 \times d_2}{d_1} = \frac{2 \text{ in.} \times 1 \text{ in.}}{10 \text{ in.}} = 0.2 \text{ in.}$$

Therefore, the clutch linkage moves 0.2 in. with a force of 50 lb.

## 5.2 ENERGY

### Energy Conversion

Energy commonly exists in four different forms; thermal (heat), mechanical (motion), electrical, and chemical (fuels). These forms of energy can be converted from one type to another using various processes or machines. The process of combustion releases chemical energy into thermal energy. The resulting heat may be used directly—say, to heat your home—or may be converted into mechanical energy, as with a steam turbine (or gasoline engine). Mechanical energy can be converted into electrical energy with a generator. Electrical energy can be converted back into mechanical energy with an electric motor and into thermal energy with resistance coils.

Thermal energy is measured in British thermal units (Btu), where 1 Btu is the amount of heat it takes to heat 1 lb of water 1°F. In SI units, thermal energy is measured in joules (J), where 1 J is the amount of heat it takes to heat 0.239 gram of water 1°C.

Mechanical energy is measured in ft · lb, where 1 ft · lb is the energy it takes to lift 1 lb vertically 1 ft. In SI units, mechanical energy is measured in N · m, where 1 N · m is the energy it takes to lift an object weighing 1 N vertically 1 m.

To find the equivalent amount of different types of energy, we can use standard conversion factors as given in Table 5.2.

Converting between energy types is simply a matter of using these conversion factors, but remember that energy and power are related but different. **Energy** is the *amount of work it takes to do a job*, and **power** is the *rate at which the energy is used*. A more powerful motor allows you to do a particular job in less time than a weaker motor, but the amount of energy expended is the same in both cases. Examples of energy-power conversion are given next.

**TABLE 5.2**

#### Energy Conversion Factors

*Chemical* → *Thermal*

1 gal fuel oil → 142,500 Btu (or  $1.5 \times 10^8$  J)

*Thermal* → *Mechanical*

1 Btu → 778 ft · lb (or 1055 N · M)

1 J (or 0.239 calories) → 1 N · m

*Mechanical* → *Electrical*

1 ft · lb →  $3.76 \times 10^{-4}$  W · h

1 N · m →  $2.78 \times 10^{-4}$  W · h

**EXAMPLE 5.7**

A small 2000-Btu/h oil-fired furnace is to be replaced by an electric furnace operating on 120 Vac. How much current will the electric furnace require?

**SOLUTION**

First, convert the thermal energy per hour into its electrical equivalent. In this case we have to work through two conversion factors to get the desired result, because we don't have one that converts directly from Btu/h to watts.

$$\frac{2000 \text{ Btu}}{\text{h}} \times \frac{778 \text{ ft} \cdot \text{lb}}{\text{Btu}} \times \frac{3.76 \times 10^{-4} \text{ W} \cdot \text{h}}{\text{ft} \cdot \text{lb}} = 584 \text{ W}$$

The result is measured in watts, which is a unit of power because the 2000 Btu/h is a "power," being energy per unit time. Knowing the voltage to be 120 V, we can find the current:

$$P = VI$$

$$I = \frac{P}{V} = \frac{584 \text{ W}}{120 \text{ V}} = 4.86 \text{ A}$$

Therefore, the electric furnace would require a current of 4.86 A.

**EXAMPLE 5.8**

A high-efficiency electric motor used on a winch draws 2 A at 24 Vdc. How fast can it lift a 100-lb weight?

**SOLUTION**

First, calculate the electric power in the motor:

$$P = VI = 24 \text{ V} \times 2 \text{ A} = 48 \text{ W}$$

Now convert electrical power into mechanical power:

$$48 \text{ W} \times \frac{1 \text{ ft} \cdot \text{lb}}{3.76 \times 10^{-4} \text{ W} \cdot \text{h}} = 127,700 \text{ ft} \cdot \text{lb/h}$$

To see how fast the motor can lift 100 lb, we divide by 100 lb (so the units of pounds cancel):

$$\frac{127,700 \text{ ft} \cdot \text{lb/h}}{100 \text{ lb}} = 1277 \text{ ft/h, or } 21.3 \text{ ft/min}$$

Therefore, the winch could lift 100 lb at a rate of 21.3 ft/min.



**EXAMPLE 5.8 (Repeated with SI Units)**

A high-efficiency electric motor used on a winch draws 2 A at 24 Vdc. How fast can it lift a 50-kg mass?

**SOLUTION**

First, calculate the electric power in the motor:

$$P = VI = 24 \text{ V} \times 2 \text{ A} = 48 \text{ W}$$

Now convert electrical power into mechanical power:

$$48 \text{ W} \times \frac{1 \text{ N} \cdot \text{m}}{2.78 \times 10^{-4} \text{ W} \cdot \text{h}} = 172,660 \text{ N} \cdot \text{m/h}$$

To see how fast the motor can lift 50 kg, we first find the force required to lift 50 kg in the presence of gravity:

$$F = ma = mg = 50 \text{ kg} \times 9.8 \text{ m/s} = 490 \text{ N}$$

Now divide the power by 490 N (so the units of Newtons cancel):

$$\frac{172,660 \text{ N} \cdot \text{m/h}}{490 \text{ N}} = 352 \text{ m/h, or } 5.9 \text{ m/min}$$

Therefore, the winch could lift 50 kg at a rate of 5.9 m/min.

Two general principles govern energy conversion: First, you can never get *more* energy out of a conversion than you put in; second, no energy conversion is perfect—there is always some waste, which is typically in the form of heat. Both concepts are expressed in the following equation:

$$\text{Energy equation: } E_{\text{in}} = E_{\text{out}} + E_{\text{waste}} \quad (5.17)$$

We account for this waste energy with the concept of efficiency. The **efficiency** of an energy conversion is the percentage of useful energy out. For example, the typical automobile engine is only about 20% efficient, meaning that only about 20% of the chemical energy available in the gasoline is converted into useful mechanical energy. The remaining 80% is lost as heat (both through the radiator and exhaust) and internal friction.

**EXAMPLE 5.7**

An electric motor operates on 120 V and draws 5 A. If it is 90% efficient, how much power is lost to heat?

**SOLUTION**

First, we compute the amount of power the motor is using:

$$\begin{aligned}\text{Power} &= VI \\ &= 120 \text{ V} \times 5 \text{ A} = 600 \text{ W}\end{aligned}$$

Using the rated efficiency, we can compute how this power is converted:

$$0.9 \times 600 \text{ W} = 540 \text{ W of mechanical power}$$

$$0.1 \times 600 \text{ W} = 60 \text{ W of waste heat}$$

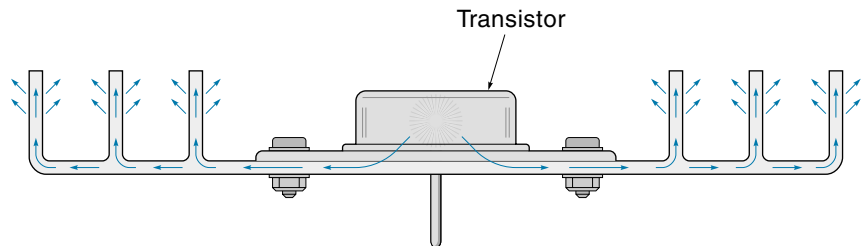
With 60 W of waste heat, this motor would give off about as much heat as a 60-W light bulb. (Incandescent bulbs convert only about 10% of their electric energy into actual light; the rest is heat.)

**Heat Transfer**

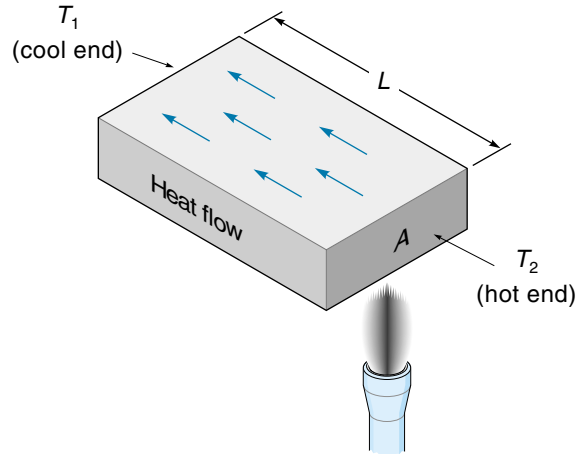
Because waste heat is generated in many electrical and mechanical systems, there must be some provision to dispose of it. In many cases, the waste heat is simply transferred to the surrounding air. For example, consider the power transistor shown in Figure 5.15. The transistor is mounted on a *heat sink*, which is a piece of aluminum with “fins” (such as found on an air-cooled engine). The fins provide a larger surface area for the heat to be transferred to the air; if the air is moving (say, from a fan), you get even better heat transfer. In Figure 5.15, the heat must travel from its source (the NPN junction) to the transistor case, to the heat-sink base, and out to the fins.

The amount of heat that conducts through a material is dependent on three factors; type of material, its shape, and the temperature difference between the ends. Figure 5.16 shows heat conducting through a rectangular solid of length  $L$  and cross-sectional area  $A$ . Heat “flows” because the temperature of the hot end ( $T_2$ ) is higher than the temperature of the cool end ( $T_1$ ), and the actual amount of that conduction can be calculated with Equation 5.18:

**Figure 5.15**  
Heat conducting  
from a transistor,  
through a heat  
sink, to the air.



**Figure 5.16**  
Heat conduction  
through a solid.



$$H = \frac{KA(T_2 - T_1)}{L} \quad (5.18)$$

where

$H$  = amount of heat being conducted

$K$  = constant for different materials

$A$  = cross-sectional area of the material

$T_2, T_1$  = high and low temperatures at each end of the material

$L$  = length of the material

Table 5.3 gives a sample of the heat conduction constant  $K$  for several materials. Table 5.3 clearly shows that metals—in particular, copper and aluminum—are the best conductors of heat. The poorest conductor is air, which explains why it is a challenge to conduct away large amounts of waste heat to the air. A simple heat-conduction problem is worked in Example 5.10.

**TABLE 5.3**

**Heat Conduction of Various Substances**

Material	$K[(\text{Btu} \cdot \text{ft})/(\text{h} \cdot \text{ft}^2 \cdot ^\circ\text{F})]$	$K[\text{cal} \cdot \text{cm}/(\text{s} \cdot \text{cm}^2 \cdot ^\circ\text{C})]$
Copper	223	0.92
Aluminum	119	0.49
Steel	29	0.12
Glass	0.48	0.002
Wood	0.048	0.0002
Cork, felt	0.024	0.0001
Air	0.014	0.000057

**EXAMPLE 5.10**

A power transistor is mounted on one end of an aluminum bracket that is 2 in. wide and  $\frac{1}{8}$  in. thick. The other end of the bracket is mounted on a large chassis (Figure 5.17). The transistor is about 1 in. from the chassis. The transistor case can be about  $100^{\circ}\text{C}$  ( $180^{\circ}\text{F}$ ) above the chassis temperature. How much power can the bracket transfer?

**SOLUTION**

We will use Equation 5.18 to calculate the heat flow, but first we need to calculate the length and cross-sectional area of the heat conductor (in feet):

$$\text{Length} = 1 \text{ in.} = 0.0833 \text{ ft}$$

$$\text{Area} = \frac{1}{8} \text{ in.} \times 2 \text{ in.} = 0.25 \text{ in.}^2 = 0.00174 \text{ ft}^2$$

Now apply Equation 5.18:

$$\begin{aligned} H &= \frac{KA(T_2 - T_1)}{L} \\ &= \frac{119 \text{ Btu} \cdot \text{ft}}{\text{h} \cdot \text{ft}^2 \cdot ^{\circ}\text{F}} \times \frac{0.00174 \text{ ft}^2 \times (180^{\circ}\text{F})}{0.0833 \text{ ft}} = 446 \text{ Btu/h} \end{aligned}$$

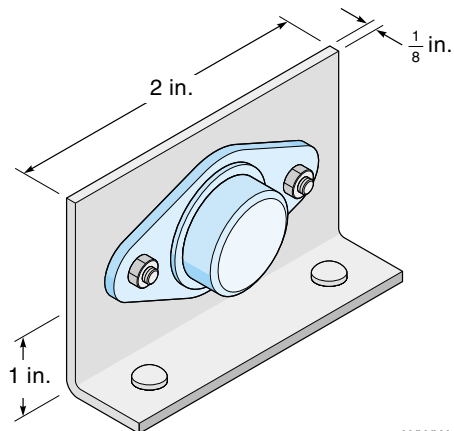
Next convert 446 Btu/h to watts using the conversion factors from Table 5.2.

$$\frac{446 \text{ Btu}}{\text{h}} \times \frac{778 \text{ ft} \cdot \text{lb}}{1 \text{ Btu}} \times \frac{3.76 \times 10^{-4} \text{ W} \cdot \text{h}}{1 \text{ ft} \cdot \text{lb}} = 130 \text{ W}$$

The aluminum bracket can transfer 130 W of thermal energy. In practice, this number would be derated to account for the thermal resistance across the transistor and chassis-mounting areas.

**Figure 5.17**

A power transistor on a bracket (Example 5.10).



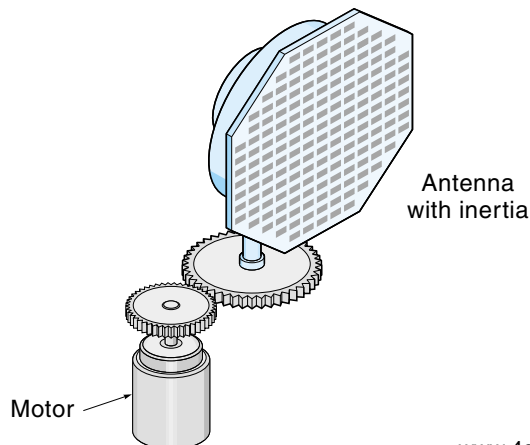
## 5.3 RESPONSE OF THE WHOLE MECHANICAL SYSTEM

Consider the case of a rotating antenna system (Figure 5.18). A motor drives a gear train, which rotates the antenna. It all seems simple enough; if the antenna needs to go to a certain position, the motor comes on, rotates until the antenna is in position, and then stops. This is *not* a simple situation, however. First, the antenna has inertia, so it takes more force to get it moving than it does to keep it moving. Second, the power train—consisting of gears, shafts, and other metal parts—has a certain flexibility. For example, if the antenna were clamped immobile and the motor tried to turn, all it could do is twist the input gear a little bit, like winding up a stiff spring. This springiness comes about because all parts in the power train will bend a little under stress. The other real-life consideration is friction. All moving parts in the system are turning on bearings, which have friction. The friction force resists motion in either direction.

### Underdamped, Critically Damped, and Overdamped Mechanical Systems

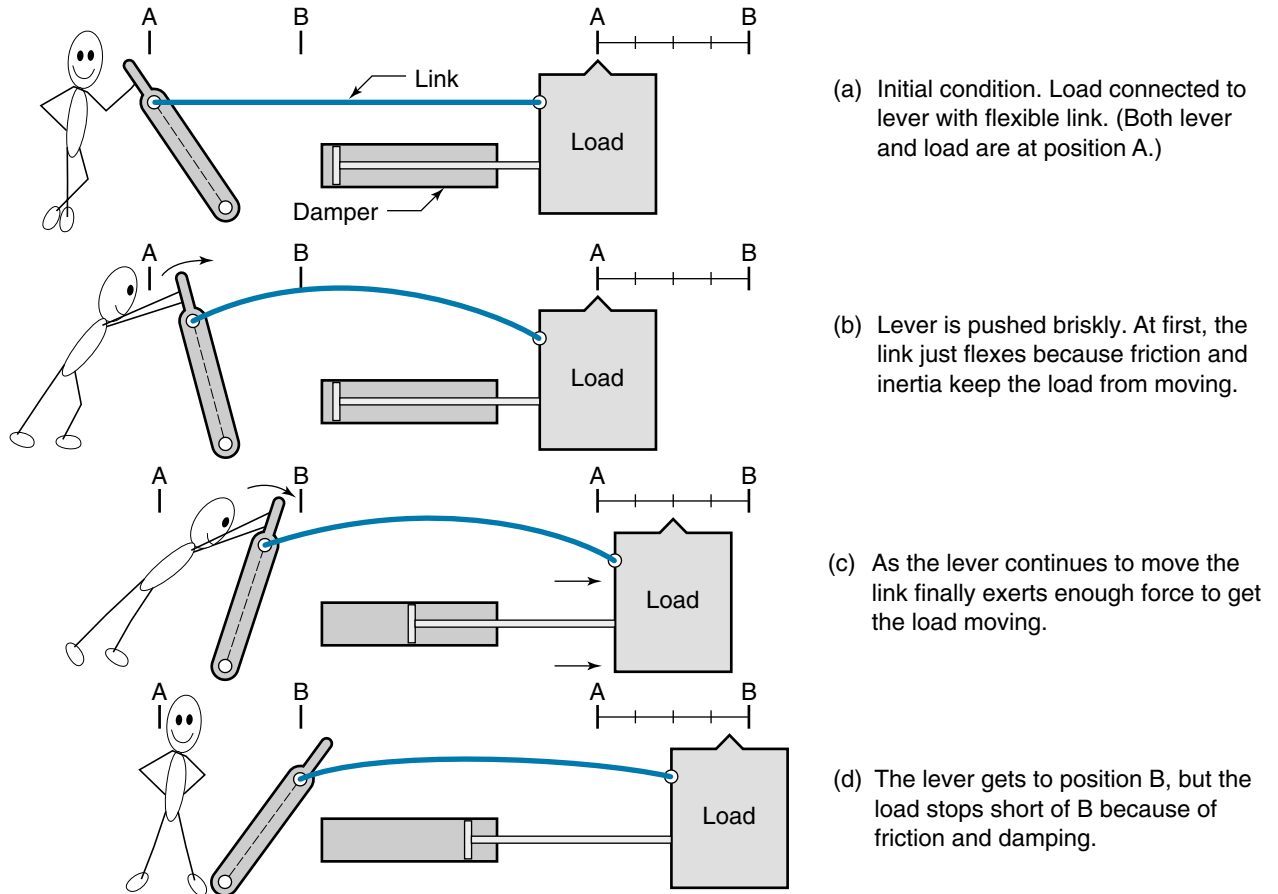
Any mechanical system, such as an antenna (see Figure 5.18), can be represented by the diagram shown in Figure 5.19. In the diagram, the load that must be moved is represented by a box. Because this load has mass (inertia for rotating systems), it resists being accelerated or decelerated. Furthermore, there is always some external drag on the load from friction and other sources (such as wind resistance or intentional hydraulic dampers). This drag is called **damping**. In Figure 5.19 the damping force is represented by a shock absorber. The link between the power source (lever) and the load is shown as

**Figure 5.18**  
A rotating  
antenna system.



**Figure 5.19**

Model of a mechanical system  
(a load being pushed by a force through a flexible link).



a flexible rod that acts like a spring. This rod represents the flexibility of the power train (flexing of levers, gears, and so on).

Note that in Figure 5.19(a) the load is at rest at position A and needs to move to position B. In Figure 5.19(b) the lever has been moved to the right (pushing on the link), but both friction and inertia prevent the load from responding immediately, so at first the link just bends. At some point, the pressure on the load from the bent link is large enough to start the load moving [Figure 5.19(c)]. Eventually, the load will come to rest somewhere near point B [Figure 5.19(d)]. The motion of a load in response to the force is said to be either **underdamped**, **critically damped**, or **overdamped** as described next.

**Figure 5.20**

The effect of damping on a mechanical system response.

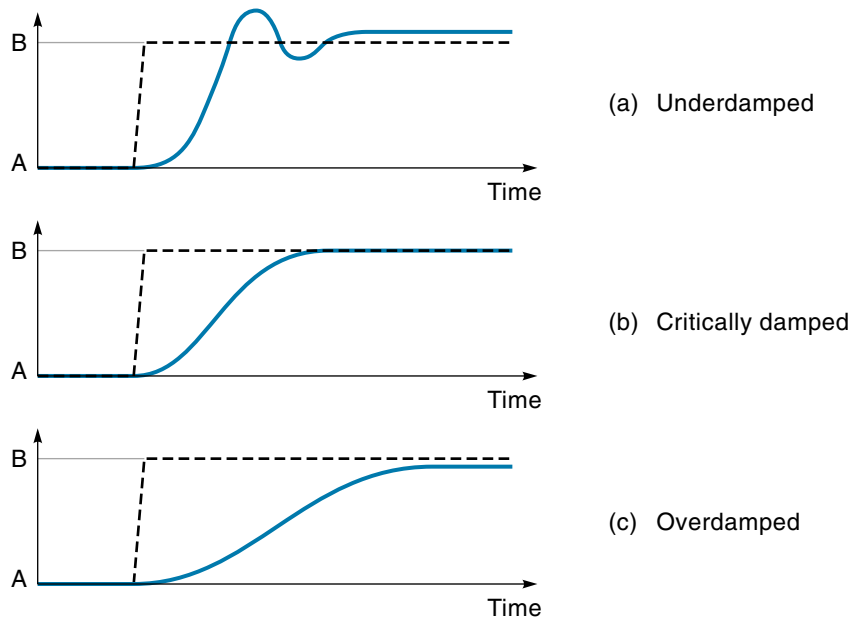


Figure 5.20(a) shows the response of an underdamped system. Recall that the load was at rest at position A, and then a lever gave it a push. The dashed line shows the movement of the lever. At first, the load lags behind a little, but if the connecting link is stiff enough (and the load is not too heavy), the load will speed up quickly. In fact, the inertia of the fast-moving load carries it past point B; this is called **overshoot**. The load may oscillate back-and-forth several times before coming to rest. You can see now why this system is called underdamped because the friction (damping force) is not enough to keep the load from overshooting its destination.

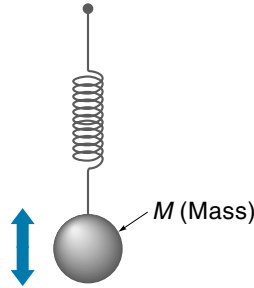
Figure 5.20(b) shows the response of a critically damped system. In this case, the damping is just enough to prevent overshoot. Notice that the time to get from position A to B is a little longer, but the absence of oscillations is desirable in many cases. In fact, the time required to actually settle out is the shortest for a critically damped system.

Figure 5.20(c) shows the response of an overdamped system. Overdamped is any damping more than critically damped, so the response time is even slower—one might say sluggish. Also notice that the final resting place of the load may fall considerably short of position B (the system shown in Figure 5.19 was overdamped, and it never did reach its new set point).

In real control systems, the controller can compensate somewhat for the natural mechanical tendencies that have been discussed here. For example, overshoot [Figure 5.20(a)] could be greatly reduced with the proper control algorithm. (These concepts are discussed in Chapter 11.)

**Figure 5.21**

Mass ( $m$ ) will oscillate up and down on a spring.



## Mechanical Resonance

Figure 5.21 shows a mass that is supported by a spring. If you pull the mass down a little and let it go, it will oscillate up-and-down. The frequency of the oscillations is called the **natural resonant frequency** and can be calculated by Equation 5.19:<sup>\*</sup>

$$f = \frac{1}{2\pi} \sqrt{\frac{k}{M}} \quad (5.19)$$

where

$f$  = resonant frequency (in Hz)

$k$  = spring force constant

$m$  = mass

With the mass and spring setup of Figure 5.21, it's easy to visualize the mass oscillating up and down. The point is that all mechanical parts have mass and all are connected to other parts through a power train or a superstructure that has some spring constant  $k$ . If all mechanical parts in a system have mass and are connected by “springs,” then all parts will have resonant frequencies. Figure 5.22 shows two examples. In Figure 5.22(a), a mass is in the middle of a steel support bracket. The bracket is the “spring.” If you push down on the mass, the bracket will bend (shown with dashed line). When you let up, the bracket will return to its normal position. This is a stiff spring, meaning the spring constant ( $k$  in Equation 5.19) will be relatively high, which in turn causes the mechanical resonant frequency to be high, perhaps 500 Hz. So although you couldn't actually see it vibrate, you could probably hear it because 500 Hz is in the audio range. The setup in Figure 5.22(b) has the same mass, but this time the support bracket is anchored at only one end. It will take much less force to bend the bracket in this configuration; thus, the spring constant is less, which (again, according to Equation 5.19) causes the resonant frequency to be lower, perhaps 25 Hz.

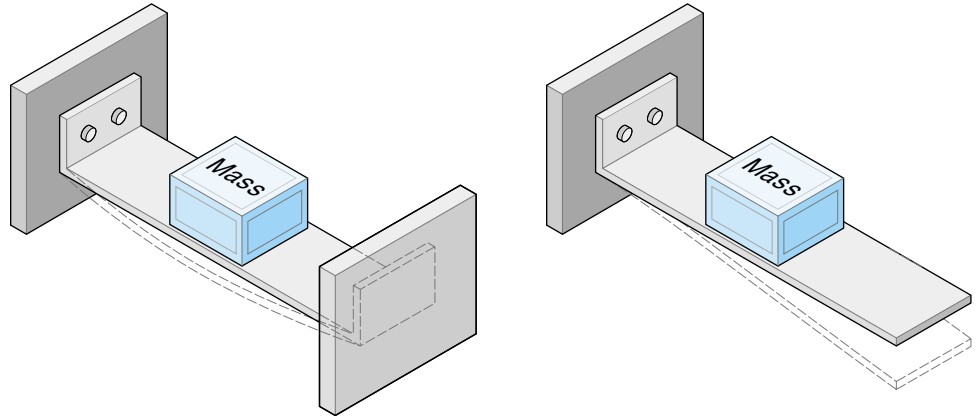
The concept of resonant frequency is important because it explains why mechanical parts will vibrate, buzz, or develop cracks and break under certain conditions. These events happen because the part is oscillating back-and-forth at its resonant frequency. It takes surprisingly little energy to keep a part vibrating *if the energy itself is applied in the form of*

<sup>\*</sup>Notice how similar this equation is to the equation for electrical resonance:  $f_c = 1/(2\pi\sqrt{LC})$ .



**Figure 5.22**

A support bracket acts as a spring:  
 (a) Stiff support.  
 (b) Support is much less stiff when support is provided on only one end.



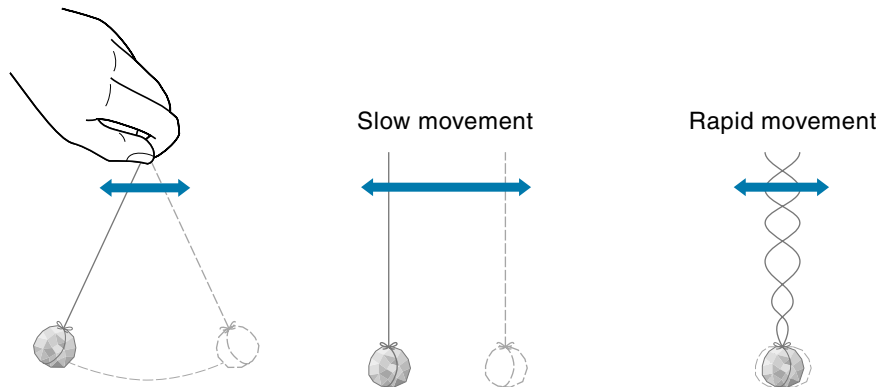
(a) Support on both ends makes a stiffer support

(b) Support on one end makes a springier support

*a vibration at or near the resonant frequency.* You can demonstrate this easily with a length of string and a mass (say, a small rock), as illustrated in Figure 5.23. First, get the rock swinging [Figure 5.23(a)] and notice how little energy it takes to keep the rock oscillating; you barely have to move your hand at all. Next slow the frequency way down so that you are slowly moving your hand back and forth [Figure 5.23(b)]. What happens to the rock? It slowly follows under your hand, with no wild oscillations. Finally, increase your hand frequency way above the resonant frequency [Figure 5.23(c)]. The rock pretty much stays still, and the string absorbs the motion. The important conclusion: A mechanical part may make wild movements if it is driven by a vibration at or near its resonant frequency.

**Figure 5.23**

A small rock and string demonstrate  
 (a) natural resonant frequency, (b) below-resonant frequency, and (c) above-resonant frequency.



(a) Large oscillations when driven at resonant frequency

(b) Minimal oscillations when driven below resonant frequency

(c) Minimal oscillations when driven above resonant frequency

Sources of mechanical vibrations abound in machinery—from motors, gears, and bearings or from *anything* that moves in a repeated motion. Mechanical engineers sometimes use special computer programs that analyze and predict the resonant frequency of each part of the mechanical design. If the design is something that must be right the first time, such as a space satellite, a prototype of the device is mounted on a test machine called a shake table, which vibrates the new design at various frequencies while high-speed cameras record the response. Any wild mechanical oscillations can then be identified and corrected.

### EXAMPLE 5.11

The electrical control box for an electric motor is attached to the motor case with a bracket (Figure 5.24). The bracket is developing cracks even though it is clearly strong enough to support the weight of the box. What is the problem here?

### SOLUTION

The likely problem is that the box is vibrating back-and-forth on the bracket when the motor is running, thus putting a strain on the bracket. That is, we suspect that the resonant frequency of the box swinging on the bracket is at or near the motor rpm. To test this theory, first determine the spring constant of the bracket by noticing that it deflects 0.1 in. when pushed with 10 lb:

$$k = \frac{\text{force}}{\text{deflection}} = \frac{10 \text{ lb}}{0.1 \text{ in.}} = 100 \frac{\text{lb}}{\text{in.}}$$

Next determine the mass of the box. It is found to weigh 1 lb. We then find the mass by dividing by the gravity term, which is 32 ft/s<sup>2</sup>, or 384 in./s<sup>2</sup>:

$$m = \frac{w}{g} = \frac{1 \text{ lb}}{384 \text{ in./s}^2} = 0.0026 \frac{\text{lb} \cdot \text{s}^2}{\text{in.}}$$

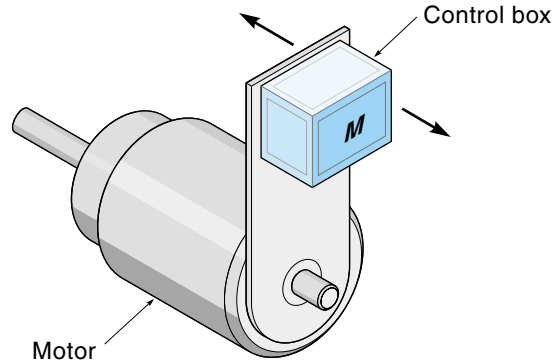
Now use Equation 5.19 to calculate the resonant frequency of the box swinging on the bracket:

$$f = \frac{1}{2\pi} \sqrt{\frac{k}{m}} = \frac{1}{6.28} \sqrt{\frac{100 \text{ lb/in.}}{0.0026 \text{ lb} \cdot \text{s}^2/\text{in.}}} = 31 \text{ Hz}$$

From the motor plate, we read that it operates at 1750 rpm, which works out to 29 rps (dividing by 60). The motor is vibrating at 29 Hz, which is very close to the resonant frequency of the box: 31 Hz. No doubt, the box has been swinging back-and-forth on the bracket. To solve the problem, we need to change the resonant frequency of the box; up or down would be sufficient. This could be done by making the bracket shorter or longer or changing its shape.

**Figure 5.24**

Control-box  
vibration on a motor  
(Example 5.11).



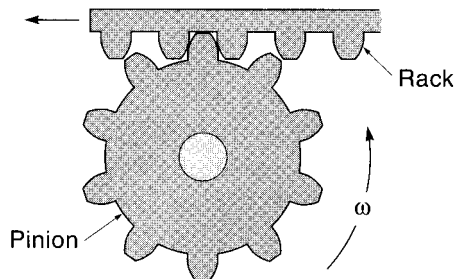
One last point on resonant frequency: If free movement of a part is restricted because of friction or some other form of damping, the magnitude of the oscillations will be reduced. In fact, if you look at Figure 5.20(a) (underdamped case), the overshoot and undershoot are the beginnings of resonant frequency oscillations, which are quickly damped out. If the part is critically damped [Figure 5.20(b)], no oscillations will occur at all—even when driven at its resonant frequency.

## 5.4 GEARS

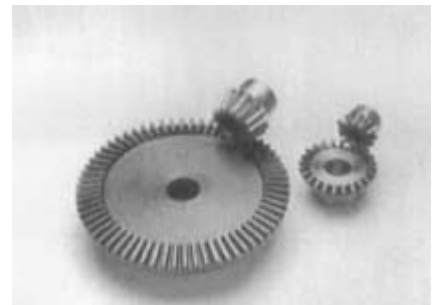
Gears allow us to change the rotational velocity and the torque to suit the motor and load conditions. Gears also can be used to simply transport power from one shaft to a parallel shaft. The concept of gearing is certainly not new. The ancient Greeks were using wooden gears for such things as transferring power from the waterwheel to the millstone (a practice still in use in existing old water-powered mills). Modern precision-cut metal and plastic gears provide an efficient, smooth, and durable means of power transmission. Figure 5.25 shows a selection of gears.

**Figure 5.25**

A selection of gear  
types. (Photos courtesy  
of Boston Gear)



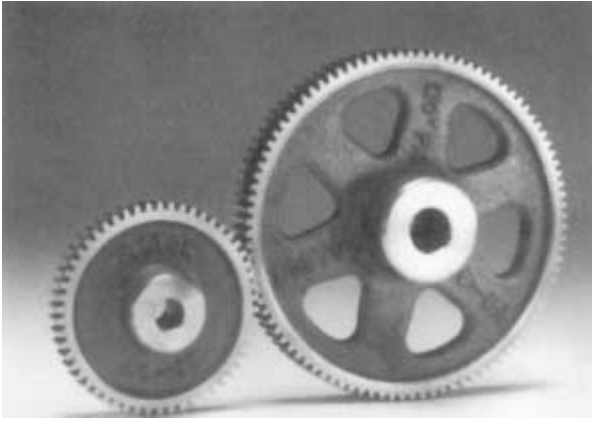
(a) Rack and pinion



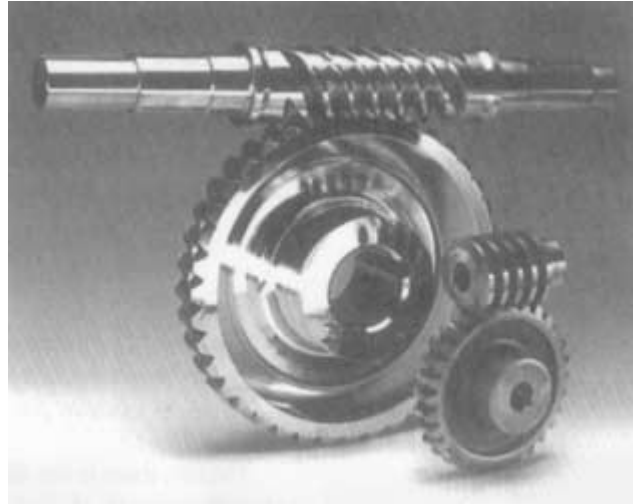
(b) Straight bevel gears

Figure 5.25

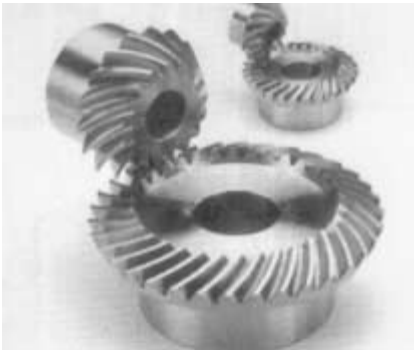
(cont.)



(c) Spur gears



(d) Worm and worm gears



(e) Spiral bevel gears

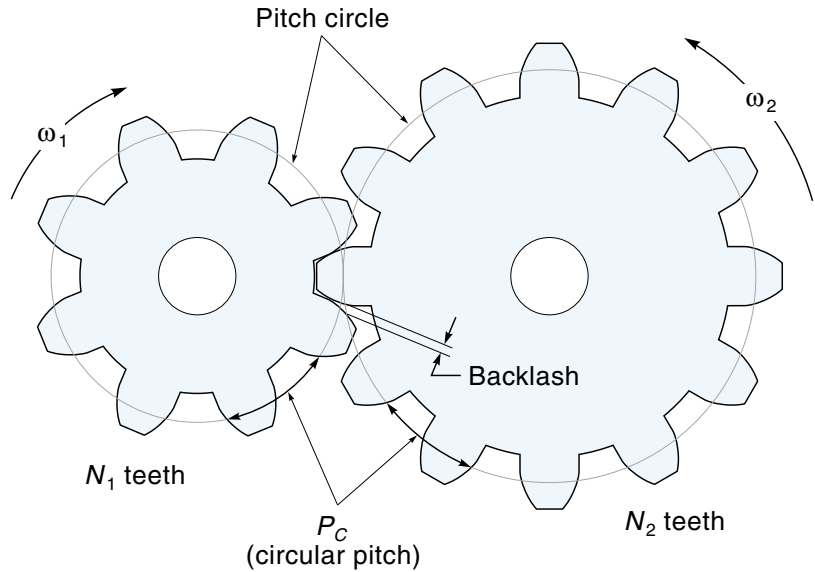
## Spur Gears

Figure 5.26 shows two meshed **spur gears**. Power is transmitted by a tooth of one gear pushing against the tooth of the mating gear, one at a time. You might think that this would cause a jerky motion; if the teeth have the proper shape, however, they actually roll on each other, handing the power smoothly over to the next tooth (Figure 5.27). Because the teeth are actually rolling and not sliding on each other, there is very little friction in a **gear pass**, which is one set of mating gears.

When two gears of different diameters are meshed, they rotate at different velocities. The motion is similar to two wheels of different diameters rolling against each other, where the size of each wheel is a theoretical circle called the **pitch circle** (see Figure 5.26). The diameter of this pitch circle is called the **pitch diameter**. Notice that the pitch diameter is smaller than the overall diameter of the gear because the teeth overlap in the mesh.

The distance along the pitch circle of one tooth (and “valley”) is called the **circular pitch**. Of course, an integer number of teeth must be

**Figure 5.26**  
Meshed spur gears.



on a gear (you cannot have a  $5\frac{1}{2}$ -tooth gear!). The number of teeth on a gear can be calculated from Equation 5.20:

$$N = \frac{\text{circumference}}{\text{distance between teeth}} = \frac{\pi D}{P_c} \quad (5.20)$$

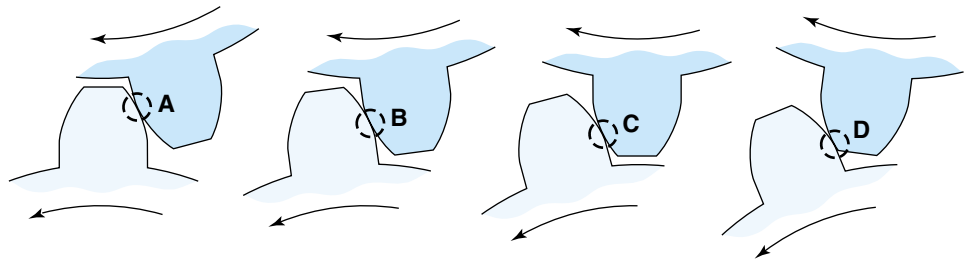
where

$N$  = total number of teeth

$D$  = pitch diameter

$P_c$  = circular pitch

**Figure 5.27**  
Showing different positions of a single involute tooth rolling on its mating tooth (follow contact point A–B–C–D).



Finally, there is the **diametral pitch**, or simply *pitch*, which is the ratio of the number of teeth per inch of pitch diameter, as shown in Equation 5.21:

$$\text{Diametral pitch} = \text{pitch} = \frac{N}{D} \quad (5.21)$$

where

$N$  = total number of teeth

$D$  = pitch diameter (in inches)

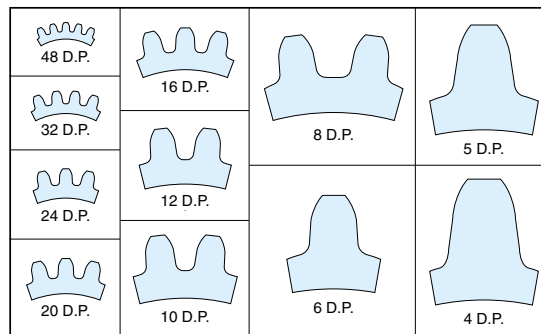
Diametral pitch is the parameter most often used to specify the tooth size in a gear. Figure 5.28 illustrates the size of teeth with pitches ranging from 4 to 48. Notice that a larger tooth has a smaller pitch. In a gear mesh, only one tooth is in contact at any time, so the larger the tooth, the more power can be transmitted. Lightly loaded machines such as printers and plotters typically use gears with a pitch between 48 and 32; these tend to be known as **instrument gears**. Automotive transmissions use a pitch range of 10-5. Of course, a bottom-line requirement is that mating gears have the same pitch (tooth size) so that they will fit together

### Using Gears to Change Speed

The **gear ratio** is a ratio of the number of teeth of two gears. The two gears in Figure 5.29 have 40 and 20 teeth, respectively, so the gear ratio is  $40/20 = 2$ . However, there is more to it. For the gears to mesh, their tooth sizes must be the same, which means they both have the same number of teeth per inch of circumference. So the gear ratio is also the ratio of circumferences. Finally, the diameter of any circle is directly proportional to its circumference, so the gear ratio is also a ratio of diameters, which makes

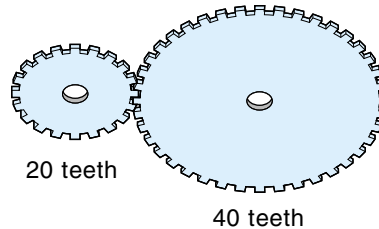
**Figure 5.28**

Gear teeth sizes for different diameter pitches. (Courtesy of Boston Gear)



**Figure 5.29**

Two gears in mesh;  
gear ratio is  $40/20 = 2$ .



it convenient to find the ratio of an existing gear mesh; all you do is measure the diameters and divide.\* This is presented in Equation 5.22:

$$N_g = \frac{N_2}{N_1} = \frac{\text{Cir}_2}{\text{Cir}_1} = \frac{\pi \text{Dia}_2}{\pi \text{Dia}_1} = \frac{\text{Dia}_2}{\text{Dia}_1} \quad (5.22)$$

where

$N_g$  = gear ratio between gears 1 and 2

$N$  = number of teeth

Cir = circumference of the gear

Dia = diameter of the gear

The assumption in Equation 5.22 is that gear 1 is supplying the power (called the driver gear), and gear 2 is receiving the power (called the driven gear).

### EXAMPLE 5.12

What is the gear ratio of the gear mesh shown in Figure 5.30?

### SOLUTION

The driver gear has a diameter of 0.75 in., and the driven gear has a diameter of 2.25 in. Find the ratio by dividing the diameters:

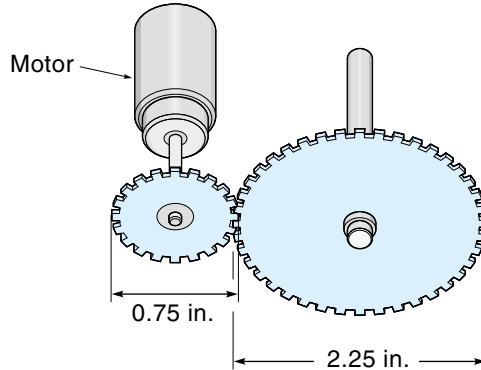
$$N_g = \frac{\text{Dia}_2}{\text{Dia}_1} = \frac{2.25 \text{ in.}}{0.75 \text{ in.}} = 3$$

Looking at the 3 : 1 gear mesh in Figure 5.30, you can see that the small gear (called the **pinion**) makes three revolutions in the same time it takes the big gear to make one revolution. In other words, the small gear must rotate three times faster than

\*Technically, you should divide the pitch diameters, but with most small-toothed instrument gears, the overall diameter and the pitch diameter are practically the same.

**Figure 5.30**

A 3:1 gear mesh  
(Example 5.12).



the big gear. This kind of logic leads us to conclude that the ratio of position, velocity, and acceleration are inversely proportional to the gear ratio  $N_g$ :

$$N_g = \frac{N_2}{N_1} = \frac{\theta_2}{\theta_1} = \frac{\omega_2}{\omega_1} = \frac{\alpha_2}{\alpha_1} \quad (5.23)$$

where

$N_g$  = gear ratio

$N$  = number of teeth

$\theta$  = gear position (angle in degrees)

$\omega$  = angular velocity of the gear

$\alpha$  = angular acceleration

In applying this equation, sometimes confusion arises about how to plug in the gear ratio; if the ratio is 3, does this go into the equation: as 3/1 or 1/3? The rule that never fails is this: Look at the gears (or a sketch); the bigger gear will always rotate *less* of an angle and at *less* velocity than the small gear. Plug in the gear ratio such that this relationship holds. The following examples illustrate the principles of Equation 5.23.

### EXAMPLE 5.13

A radio tuner is connected to the tuning knob through a 3 : 1 gear mesh (Figure 5.31). If the knob is turned  $70^\circ$ , how many degrees does the tuner rotate?

### SOLUTION

The gear attached to the tuner is the bigger of the two; therefore, it must turn *less* than  $70^\circ$ . We will use the part of Equation 5.23 that deals with position, solving for the angle of the tuner gear,  $\theta_2$ :

$$N_g = \frac{3}{1} = \frac{70^\circ}{\theta_2}$$



so

$$\theta_2 = \frac{70^\circ}{3} = 23.33^\circ$$

The tuner will rotate  $23.33^\circ$  when the knob rotates  $70^\circ$ .

### EXAMPLE 5.14

A small motor running at 200 rpm drives a paper roller in a business machine (Figure 5.32). The gear on the motor has 20 teeth, and the gear on the roller has 50 teeth. How fast is the roller turning?

### SOLUTION

First, note that the roller gear is larger than the motor gear, so it will be turning slower than the motor. To calculate the speed, apply Equation 5.23, solving for velocity:

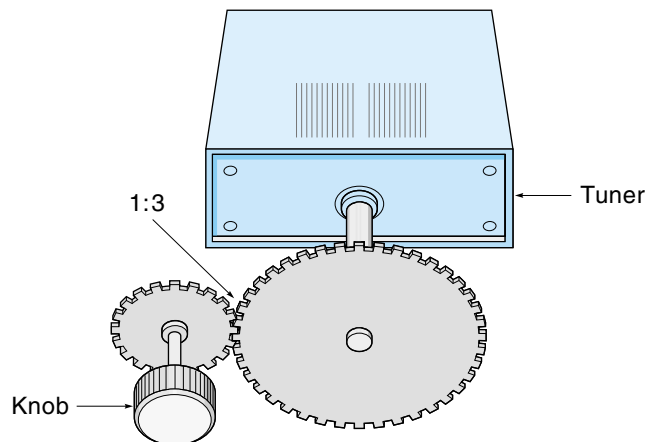
$$\frac{50 \text{ teeth}}{20 \text{ teeth}} = \frac{200 \text{ rpm}}{\omega_2}$$

so

$$\omega_2 = \frac{200 \text{ rpm} \times 20 \text{ teeth}}{50 \text{ teeth}} = 80 \text{ rpm}$$

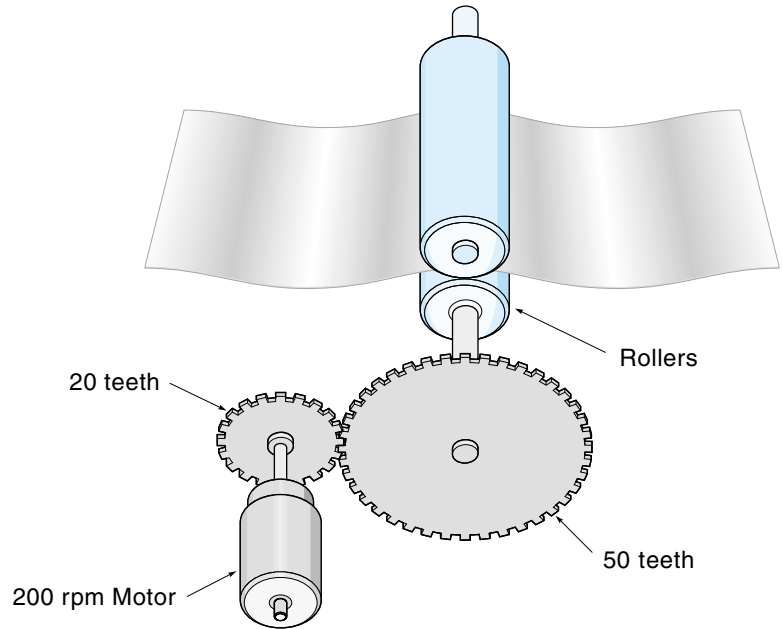
**Figure 5.31**

A gear setup  
(Example 5.13).



**Figure 5.32**

Gear drive for a paper roller (Example 5.14).



A **gear train** consists of more than a single gear pass. Figure 5.33 shows a system with three gears. Notice that the middle gear is actually two gears fastened together. The gear ratio of the total system is the product of the individual gear ratios:

$$N_{g(\text{tot})} = N_{g1} N_{g2} N_{g3} \dots \quad (5.24)$$

where

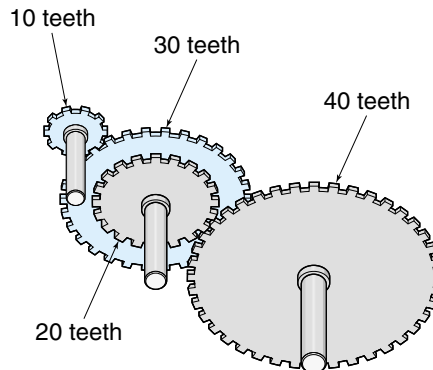
$N_{g(\text{tot})}$  = overall gear ratio

$N_{g1}$  = gear ratio of first pass

$N_{g2}$  = gear ratio of second pass, and so on

**Figure 5.33**

A two-pass gear train.



**EXAMPLE 5.15**

Find the overall gear ratio of the gear train pictured in Figure 5.33.

**SOLUTION**

There are actually four gears here (10, 20, 30, and 40 teeth) and two gear passes. The 20- and 30-tooth gears are fastened together and so rotate at the same velocity. The overall gear ratio is found by multiplying the two individual gear ratios:

$$N_{g1} = \frac{30 \text{ teeth}}{10 \text{ teeth}} = 3 \quad \text{Ratio of first pass}$$

$$N_{g2} = \frac{40 \text{ teeth}}{20 \text{ teeth}} = 2 \quad \text{Ratio of second pass}$$

$$N_{g1}N_{g2} = 3 \times 2 = 6 \quad \text{Overall ratio}$$

**Using Gears to Transfer Power**

For precision-made spur gears, the assumption is that power is conserved across the gear pass. Except for a small loss through friction, the gear pass can neither create nor destroy power:

$$\text{Power in, gear 1} = \text{Power out, gear 2} \quad (5.25)$$

Therefore, because power = torque  $\times$  angular velocity, we rewrite Equation 5.25 as follows:

$$T_1\omega_1 = T_2\omega_2 \quad (5.26)$$

where

$T$  = torque (such as ft  $\cdot$  lb or N  $\cdot$  m)

$\omega$  = angular velocity (such as deg/s)

Rearranging Equation 5.26, we get

$$\frac{T_2}{T_1} = \frac{\omega_1}{\omega_2} = N_g \quad (5.27)$$

Equation 5.26 tells us that the product of torque and velocity is the same on each side of the gear pass. This means that the faster turning gear has less torque, and the

gear that is turning slower has more torque.\* This is demonstrated every time you drive a car; the lower gears give more torque for start up, and the higher gears give more speed but less torque. Equation 5.27 shows that the ratio of torques in a gear pass is inversely proportional to the gear ratio.

### EXAMPLE 5.16

An electric motor supplies 60 in. · oz of torque while running at 100 rpm and it is driving a load through a 1 : 5 gear ratio (Figure 5.34). Find the output torque and velocity.

### SOLUTION

Equation 5.27 relates torque and velocity to gear ratio. Solving first for torque on the driven (or load) gear, we have  $T_2/T_1 = N_g$ , so

$$T_2 = T_1 N_g = 60 \text{ in.} \cdot \text{oz} \times 5 = 300 \text{ in.} \cdot \text{oz}$$

Solving for the velocity of the load gear we have  $\omega_1/\omega_2 = N_g$ , so

$$\omega_2 = \frac{\omega_1}{N_g} = \frac{100 \text{ rpm}}{5} = 20 \text{ rpm}$$

This is a typical application for a gear pass. Electric motors tend to be high-speed and low-torque, whereas many applications require higher torques and lower speeds. In this case, the torque was increased by a factor of 5, to 300 in. · oz, but the speed was reduced by a factor of 5, to 20 rpm.

Recall from Section 5.1 that every load rotating on an axis has a moment of inertia ( $I$ ). The more  $I$  it has, the more energy it takes to get it spinning. Ultimately, the motor has to supply this energy; thus, the question is, What moment of inertia does the motor “see” if the load is driven through a set of gears (Figure 5.35)? It turns out that the reflected moment of inertia is inversely proportional to the gear ratio squared, as given in Equation 5.28:

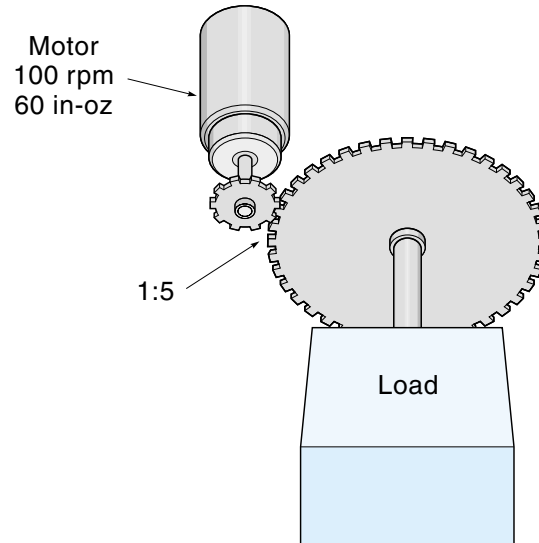
$$I \text{ reflected} = \frac{I_{\text{load}}}{N_g^2} \quad (5.28)$$

This means that in most cases the motor “sees” a much lower inertia than the load actually has. The comprehensive motor example at the end of Chapter 7 gives a practical application of the reflected inertia equation.

\*Note the similarity between a gear pass and an ideal electrical transformer, where the product of  $V \times I$  is maintained from primary to secondary windings.

**Figure 5.34**

A motor driving a load (Example 5.16).

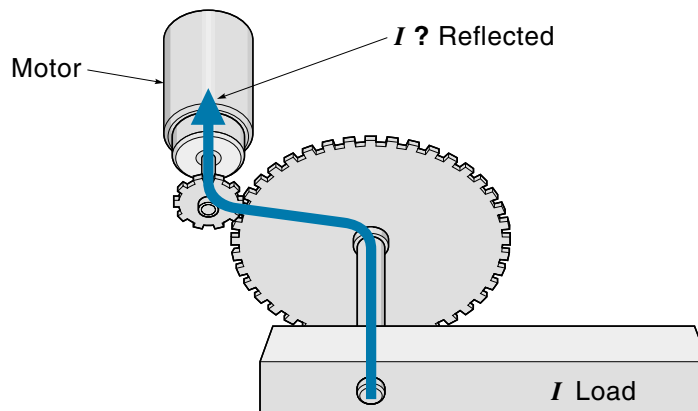


### Long Gear Trains

Large gear reductions (ratios) can be made with multiple gear passes, but this approach presents special problems. For example, a reduction of 1 : 256 might take as many as four passes using conventional spur gears. Such a gear train is shown in Figure 5.36. Undesirable qualities such as inefficiency and backlash tend to magnify with each gear pass. **Backlash** is the small free clearance between mating teeth. A small amount of backlash is necessary to prevent binding of the gears if they are slightly out of round. However, in a long gear train, the effect of the backlash is amplified along with the gear

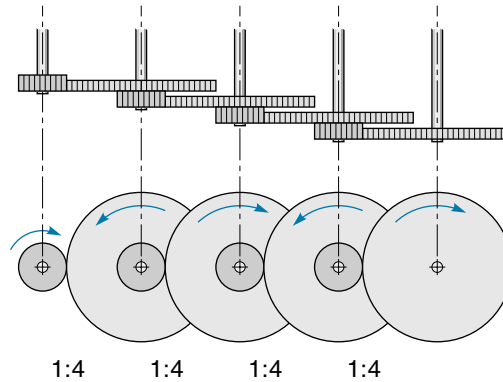
**Figure 5.35**

The moment of inertia of a load is “reflected” to the motor.



**Figure 5.36**

A gear train with four passes; total gear ratio is 256.



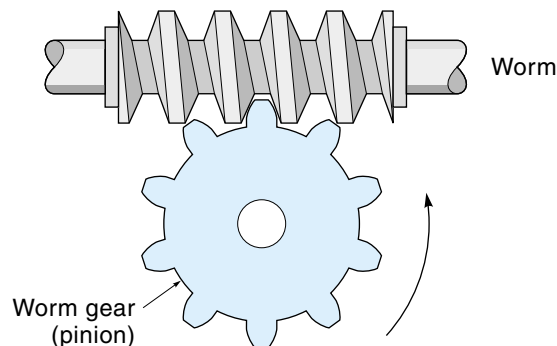
ratio. In the case of Figure 5.36, backlash of  $1^\circ$  in the first pass would be amplified by the three other gear passes ( $4 \times 4 \times 4 = 64$ ) to become  $64^\circ$  of backlash at the end—clearly unacceptable. Backlash comes into play when the direction of rotation is reversed. The driver gears have to move a bit before they make contact with the driven teeth. This makes precision positioning difficult when it is necessary to “tweak” into position by going forward, reverse, forward, reverse, etc. Too much backlash in a feedback control system may cause it to “chatter” or worse (see Chapter 11 for a discussion of stability). Efficiency is another problem. If the efficiency of each pass in Figure 5.36 was a respectable 96%, the overall efficiency for four passes would be  $96\% \times 96\% \times 96\% \times 96\% = 85\%$ .

## Worm Gears

Worm gears yield very high gear ratios on a single pass with little or no backlash, but they can be highly inefficient. Figure 5.37 shows a worm gear mesh and consists of the **worm** and the **worm gear**, or pinion. For each revolution of the worm, the pinion advances one tooth. This is what makes the high gear ratios possible; a pinion with 50 teeth would yield a gear ratio of 50. Notice, however, that as the worm rotates it is really

**Figure 5.37**

Worm gearing.



sliding across the pinion's teeth. The friction from this sliding is the cause of the inefficiency, which is typically in the 50-80% range.

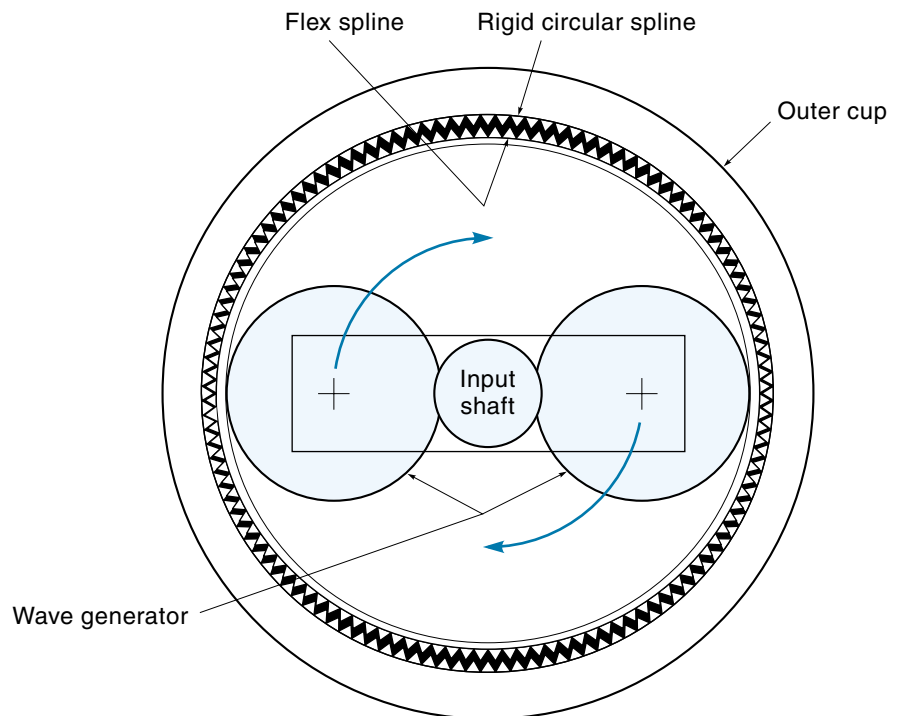
Another advantage of worm gearing is the **lockup property**. Lockup occurs in higher-ratio worm gears, and it means that power flows only one way—from the worm to the pinion. If you try to turn the pinion directly, the friction is so great that it locks up. A common use of this feature is the little worm gears used to tighten guitar strings. You can twist the worm and tighten the string, but the tension of the string cannot spin the worm gears backward.

### Harmonic Drive

The **harmonic drive** is a unique gear system that gives high efficiency, no backlash, and high gear ratio. Illustrated in Figure 5.38, the drive looks like a cup with the input shaft going through the center. The rigid cup has an internal set of teeth all the way around known as the *rigid circular spline*. Inside the rigid cup is a slightly smaller flexible cup called the *flex spline*. The flex spline typically has two teeth less than the outer cup. A rotating device called the *wave generator* pushes out on the flex spline so that the teeth mesh only on opposite sides. The input shaft is connected to the wave generator, and as it rotates, the two splines (inner and outer) mesh with each other. Because they have a

**Figure 5.38**

The harmonic drive.



different number of teeth, the outer cup (which is connected to the output shaft) will advance a little for each revolution of the flex spline. The actual gear ratio can be calculated from the following equation:

$$N_h = \frac{\text{number of teeth on outer rigid cup}}{\text{difference in number of teeth on splines}} \quad (5.29)$$

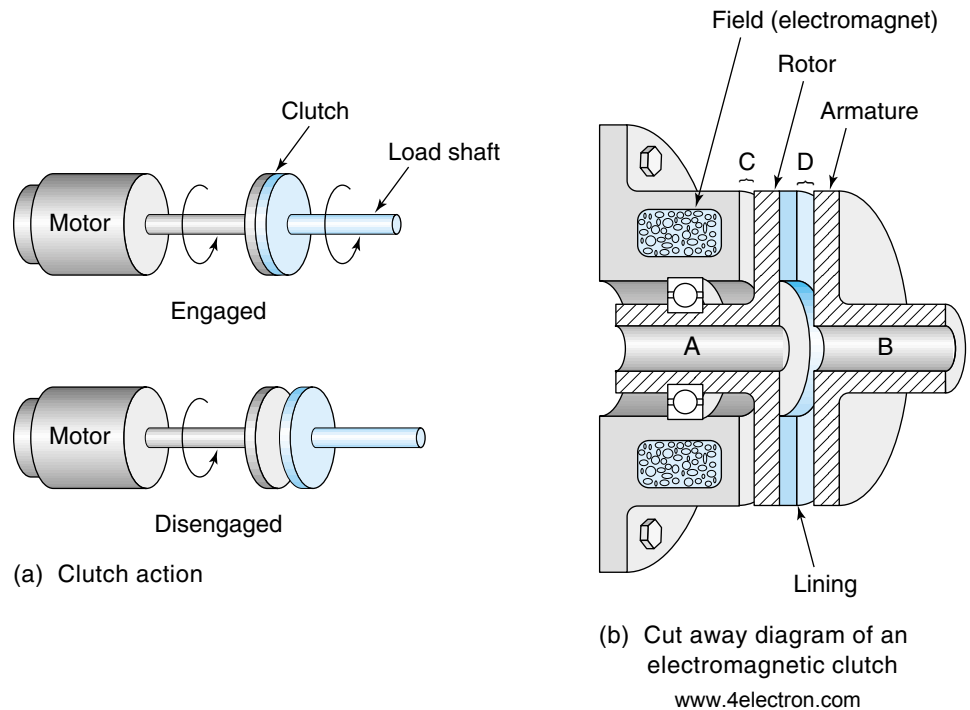
where  $N_h$  is the gear ratio (more correctly *velocity ratio*) of the harmonic drive. For example, in the drive illustrated in Figure 5.38, the number of teeth in the outer cup is 100, and the number of teeth on the flex spline is 98; thus, the gear ratio is  $100/2 = 50$ .

## 5.5 CLUTCHES AND BRAKES

### Clutches

The purpose of a **clutch** is to connect or disconnect one rotating shaft from another “on the fly.” Clutches are used in applications ranging from copy machines to factory automation to cars. Typically, one side of the clutch is connected to the motor and the other side to the load. When the clutch is disengaged, the motor can turn without driving the load. When the clutch is engaged, the motor and load shaft turn as one [see Figure 5.39(a)].

**Figure 5.39**  
Clutches





A clutch usually consists of two opposing friction disks, one attached to the motor shaft and one attached to the load shaft. The disks are pressed together to engage the clutch, and friction transmits the torque. This design allows for slippage, so the load can be brought up to speed slowly to minimize stress. Also, one of the disk surfaces may be lined with a friction material, such as asbestos, that is designed to wear and can be replaced.

Most clutches used in industry are activated by an electromagnet or with pneumatics/hydraulics. A cut-away drawing of an electromagnetic friction clutch is shown in Figure 5.39(b). It consists of three basic parts: the *field* (electromagnet), the *rotor*, and the *armature*. In the diagram, the two shafts to be “clutched” would be slid in and keyed to hubs A and B. The field coil is a large donut-shaped electromagnet in a housing. This housing is bolted to some stationary member and does not rotate. Next to the field is the disk-shaped rotor, which is mounted on bearings so that it can rotate (a small gap (C) between the field and the rotor allows the movement). Next to the rotor is the armature. In the rest state (disengaged), there is a gap (D) between the rotor and the armature so that it is free to rotate independently of the rotor. When power is applied to the electromagnet, the magnetic field jumps across gap C, goes through the rotor, and jumps across gap D to the armature. Being made of ferric (iron-based) material, the armature is pulled in tight by the magnetic field, clamping the friction faces of the rotor and armature. The clamping pressure is directly proportional to the current in the electromagnet, which makes it possible for a control system to control the slipping torque of the clutch. Pneumatic- and hydraulic-activated clutches differ mainly in whether air or fluid pressure is used to push the disks together, as explained in the next section.

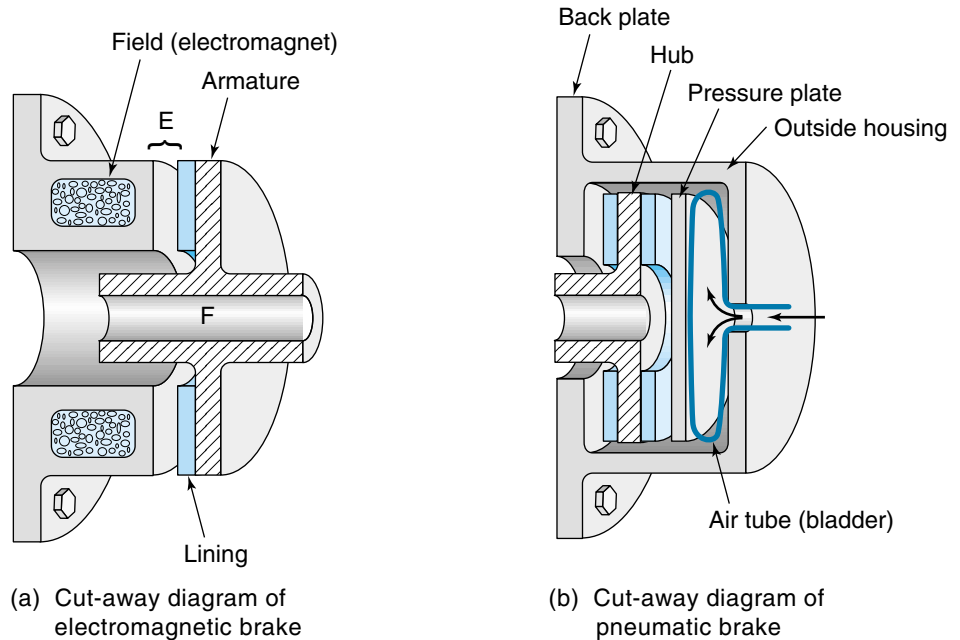
## Brakes

The purpose of a **brake** is to slow down or stop a rotating shaft, or to prevent a shaft from turning when the power has been turned off—for example, on a loaded winch. Although there are a number of brake designs, the most common is similar to that of the clutch shown in Figure 5.39(a), the difference being that in a brake, one of the friction disks is held immovable. When the brake is applied, the rotating disk is clamped to the stationary disk, thus stopping the load. Brakes can be actuated with an electromagnet, pneumatics/hydraulics, or plain brute force.

An electromagnetic brake is shown in Figure 5.40(a). It consists of two basic parts: the *field* (electromagnet) and the *armature*. Like the clutch, the field coil is an electromagnet in a housing bolted to some immovable member. Next to the field, but separated by a gap (E), is a disk-shaped armature. The armature is basically an iron disk with a hub, and it can rotate freely when the brake is not applied. The shaft to be “braked” would be slid in and keyed to hub F. When power is applied to the electromagnet, the magnetic field jumps across gap E to the armature and pulls it in tight, clamping together the friction faces of the stationary field housing and armature, thus stopping the shaft.

Figure 5.40

Brakes



Pneumatic- and hydraulic-activated brakes work in the same general way. They differ mainly in whether air or fluid pressure is used to push the disks together. Figure 5.40(b) is a cutaway diagram of a pneumatic brake. The *back plate* and *outside housing* are bolted to a stationary member and do not rotate. In the unactivated state, shaft and *hub* can turn freely inside the housing. To apply the brake, air pressure is fed through the housing into a bladder called the *air tube*. The air tube expands, pushing on the *pressure plate*, which then squeezes the friction surfaces of the hub to the back plate.

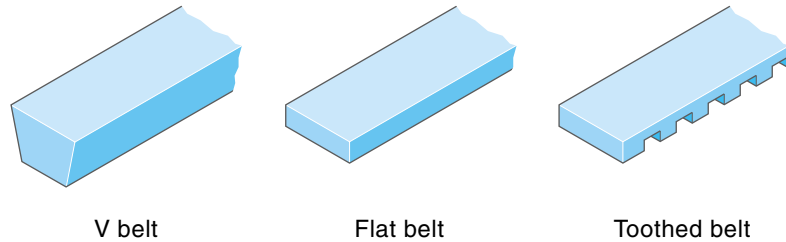
Clutches and brakes are often used together to create the rapid-cycle repeating motion found in many automated production processes. When a clutch/brake combination unit is placed between the motor and the mechanism, the motor can continue to run at a fixed speed, and the clutch/brake can be cycled on and off to cause the repeating action.

## 5.6 OTHER POWER-TRANSMITTING TECHNIQUES

The most direct way to transmit rotary power is through a shaft, such as a drive shaft or an axle shaft. Gears can transmit power between parallel or perpendicular shafts and change the velocity and torque in the process. In this section, we examine some other techniques for transmitting power from shaft to shaft.

**Figure 5.41**

Different types of power-transmission belts.



## Belts

Under the correct circumstances, belts are a good means of transmitting power. Usually made of rubber and hence very flexible, they come in various types (Figure 5.41). The advantages of belts include low cost, quiet running, low maintenance (no lubrication), shock absorption, and tolerance of nonparallel shafts, and they allow for “gearing” by having one pulley larger than the other.

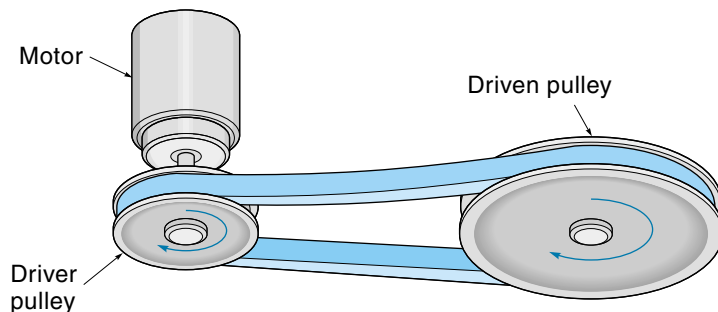
Figure 5.42 shows a standard belt setup, which includes a driver pulley and a driven pulley. Notice that the power is carried entirely by the lower belt segment. The ratio of diameters of the pulleys provides exactly the same change in velocity and torque as for gears. In fact, most equations given for spur gears (5.22–5.28) apply to belts. One difference is that meshed gears are rotating in opposite directions, whereas both pulleys in a belt system rotate in the same direction (compare Figure 5.26 with Figure 5.42).

### EXAMPLE 5.17

For the belt system shown in Figure 5.42, the diameter of the small pulley is 3 in., and the diameter of the large pulley is 8 in. The motor is rotating at 1000 rpm, providing a torque of 10 in. · lb. Find the velocity and available torque of the large pulley shaft.

**Figure 5.42**

Power transmission using a belt (Example 5.17).



**SOLUTION**

The velocity ratio  $N_v$  (similar to gear ratio) is the ratio of their diameters:

$$N_v = \frac{\omega_1}{\omega_2} = \frac{\text{Dia}_2}{\text{Dia}_1} = \frac{8 \text{ in.}}{3 \text{ in.}} = 2.67 \quad (5.30)$$

This means the big pulley will rotate 2.67 times slower than the small pulley. We can solve this problem by simply dividing 1000 rpm by 2.67 to get 375 rpm, or we can do it more formally by equations. From equation 5.27,  $N_v = \omega_1/\omega_2$ , where  $\omega_1$  and  $\omega_2$  are pulley rpms. Solving for  $\omega_2$ , we get

$$\omega_2 = \frac{\omega_1}{N_v} = \frac{1000 \text{ rpm}}{2.67} = 375 \text{ rpm}$$

The torque of the large pulley will be 2.67 larger than the small pulley. Again, we could find the output torque by simply multiplying 10 in. · lb by 2.67 to get 26.7 in. · lb, or do it by Equation 5.27 (which also applies to belts):

$$N_v = \frac{T_2}{T_1}$$

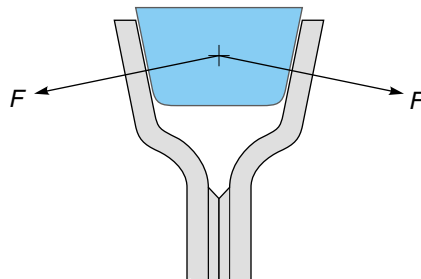
where  $T_1$  and  $T_2$  are torques. Solving for  $T_2$ , we get

$$T_2 = T_1 N_v = 10 \text{ in.} \cdot \text{lb} \times 2.67 = 26.7 \text{ in.} \cdot \text{lb}$$

With the exception of the toothed variety, belts rely on friction to transmit power. As discussed in Section 5.1, friction is proportional to how tightly the parts are pressed together. For belts, this is usually accomplished by keeping the belt in tension. A V belt increases the friction by using a wedging action (Figure 5.43). Just as a wedge produces large forces when splitting a log, the V belt produces large sideways forces that increase its grip on the pulley. Even so, a general rule is that a belt should have at least

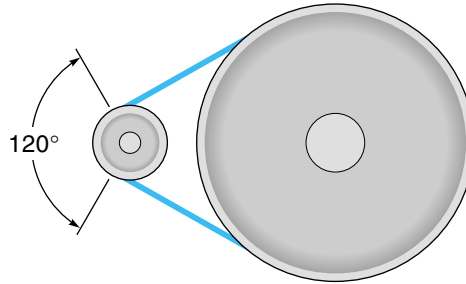
**Figure 5.43**

Belt-wedging action produces large sideways forces, increasing friction.



**Figure 5.44**

Minimum pulley contact is  $120^\circ$ .



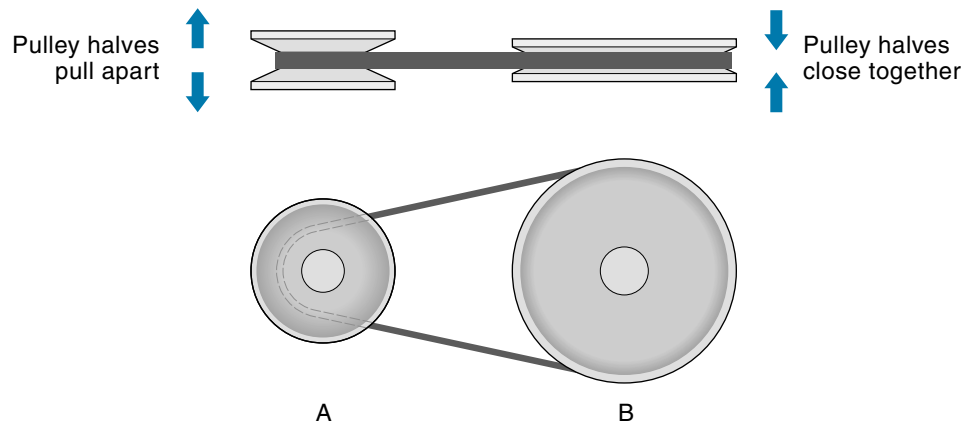
$120^\circ$  of contact to give it enough gripping area (Figure 5.44). This requirement prevents really large velocity ratios.

Untoothed belts tend to **creep** over time and cannot be used to maintain position relationships. Creep is *not* the same as slipping; slipping occurs when the pulley rotates under the belt, causing a squealing noise. Creep and slip occur because there is no positive interlocking between belt and pulley, a fact that allows for some special capabilities such as variable-speed transmission and belt clutches.

Figure 5.45 shows the variable-speed transmission. It consists of two specially made pulleys, designed so the sides can go in and out, that can effectively change their diameter while the belt is running. When the sides are close together, the belt is forced to the outside, thus making a larger diameter; when the pulley sides move apart, the belt drops down to a smaller diameter. If both pulleys (A and B) change their effective diameters at the same time but in opposite directions, then the belt stays tight even while the velocity ratio between the pulleys is changing. This is a mechanical engineer's dream: an infinite ratio transmission that can be changed "on the fly." This design is used in industry, for example, where feed speeds need to be exactly matched to other machines.

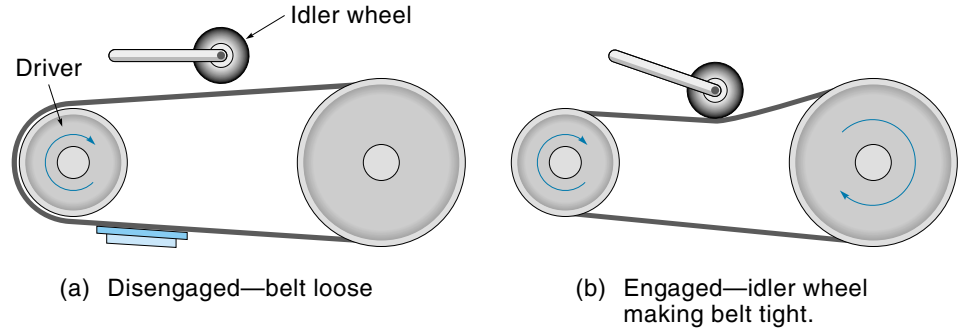
**Figure 5.45**

A variable-speed belt drive.



**Figure 5.46**

A belt clutch.

**Figure 5.47**

A roller chain.



A belt clutch is shown in Figure 5.46. The belt around the two pulleys is intentionally loose so that the driver pulley will slip when the idler wheel is up [Figure 5.46(a)]. When the idler wheel is lowered [Figure 5.46(b)], the belt tightens around both pulleys, and power is transmitted.

### Roller Chain

Another way that rotary power is transmitted is with roller chain, the kind used on bicycles. Shown in Figure 5.47, it consists of a series of links connected by pins, with a roller around each pin. The rollers engage teeth on the sprocket, so unlike belts, there is no slippage, and the “gear ratio” is a function of the number of teeth on each sprocket. Chain drives do not rely on friction as belts do, so there is no need to have the chain tight; in fact, there should always be some small slack to ensure there will be no binding. This means there will probably be backlash, which is why chains are seldom used for “back and forth” movement. Like belts, all the load is carried by one side of the chain, and both sprockets rotate in the same direction. Properly lubricated roller chain drives can be very efficient.

## SUMMARY

Most control systems include some mechanical components, so it is important to have some idea of how these components behave. Friction is the drag force that is always

present when parts slide against each other. There are three classifications of friction: static (moving an object from rest), sliding (object is moving), and viscous (sliding over a thick lubricant).

Springs come in many sizes and shapes and find many applications in mechanical systems. All springs obey Hooke's law, which states that the deformation of a spring is proportional to the applied force.

Physical objects have mass, which means they tend to resist being moved. Two basic sets of equations (from physics) relate how an object responds to a force: one for linear systems and one for rotational systems. With the appropriate givens, these equations can be used to calculate the position, velocity, or acceleration of an object.

A lever is a simple device that can amplify a physical force. A lever consists of a bar and a fulcrum, or pivot point. A weaker force applied to one end is converted into a stronger force at the other end, but the stronger force moves through a shorter distance.

Energy is usually in one of four forms: chemical (fuels), thermal (heat), mechanical (motion), and electrical. Energy can be converted from one form to another, but the efficiency of the conversion is not 100% and the "waste energy" is usually in the form of heat, as when an electric motor heats up. Managing the waste heat requires knowing something about heat conduction, which is the study of how heat moves through materials.

A real mechanical system is made up of parts, each of which has some flexibility, some inertia, and probably some frictional resistance to movement. The response of a system to a force is described as being either underdamped (too flexible), critically damped (just right), or overdamped (sluggish). Any mechanical structure has a natural resonant frequency at which it will tend to vibrate. Care should be taken to ensure that an external exciting force does not have the same frequency as the mechanical resonant frequency, or wild oscillations and breakage could result.

Gears, available in different sizes and types, are a very efficient way to transport and reconfigure rotational power. The gear ratio is the ratio of the effective gear diameters in a gear pass. The rotational velocity across a gear pass will be increased or decreased by the gear ratio. Also important is that power is conserved across the gear pass, so if the velocity increases (across a gear pass), the torque decreases. Large gear ratios can be built from multiple gear passes, called gear trains.

Clutches are used to connect or disconnect rotating shafts "on the fly." Typically, clutches are activated by electromagnetic or pneumatic/hydraulic force. Brakes are used to slow, stop, or hold a shaft. Brake construction and clutch construction are similar.

Two other power-transmitting devices are belts and roller chains. Belts have the advantage of running smooth and quiet and are tolerant to slight misalignments. They also allow for interesting designs such as clutches and variable-speed transmissions. Roller chains are more rigid than belts and can be a very efficient way to transport power.

## GLOSSARY

**acceleration** The process of increasing velocity; uniform acceleration is when the velocity is increasing at a constant rate.

**backlash** In a gear pass, the small amount of free play between gears so that the teeth don't bind on each other.

**brake** A device used to slow down or stop a shaft, or to prevent a stopped shaft from moving.

**circular pitch** The distance along the pitch circle of a gear of one tooth and valley.

**clutch** A device used to connect or disconnect rotating shafts "on the fly"; usually a clutch acts by pressing two friction disks together.

**coefficient of friction** A constant that is arrived at experimentally for different materials and can be used to calculate the force needed to overcome friction.

**creep** The property that explains why two pulleys connected with a belt will not stay exactly synchronized.

**critically damped** A system that is damped just enough to prevent overshoot.

**damping** A drag force on a system from sliding or viscous friction, which acts to slow movement (make it sluggish).

**diametral pitch** The ratio of the number of gear teeth per inch of pitch diameter; in practice, the number that describes the tooth size when specifying gears.

**efficiency** When describing energy conversions, the percentage of input energy that is converted to useful output energy.

**energy** The amount of work it takes to do a job; energy has different units for chemical, thermal, mechanical, and electrical systems.

**gear pass** Two gears in mesh.

**gear ratio** The ratio of the number of teeth of two gears in mesh. (Also the ratio of pitch diameters.)

**gear train** A gear system consisting of more than one gear pass.

**harmonic drive** A unique device using a flexible gear that can provide a large gear ratio with virtually no backlash.

**Hooke's law** The "spring law" that states the amount a spring deflects is proportional to applied force.

**inertia** The property that explains why an object in motion will tend to stay in motion. Inertia is directly related to mass; the more mass an object has, the more energy it takes to get it moving or to stop it.



**instrument gears** Smaller gears with pitch in the 48-28 range, found in office machines and smaller industrial machines.

**lever** A rigid bar pivoted at one end or somewhere in the middle and is used to amplify force much as a pry bar.

**lockup property** The property of high-ratio worm gears that cannot be driven backward.

**mass** The amount of material in an object; mass is related to weight in that the more mass it has, the more it will weigh.

**moment of inertia** A mechanical property of an object that is based on its shape and mass and the axis of rotation; the larger the moment of inertia, the more torque it takes to spin the object about the designated axis.

**momentum** The property of a moving object that tends to keep it moving in the same direction.

**natural resonant frequency** In mechanical systems, the frequency at which a part or parts will vibrate. The resonant frequency is a function of the mass and spring constant.

**Newton** Unit of force in the SI system ( $1 \text{ N} = 0.224 \text{ lb}$ )

**normal force** In friction calculations, the force pushing the sliding surfaces together.

**overdamped** A system that has so much drag (from static or viscous friction) that its response is sluggish.

**overshoot** The event when a mechanical (or electrical) output approaches its destination too fast and goes beyond; underdamped systems tend to overshoot.

**pinion** The small driver gear in a gear pass.

**pitch circle** If gears were solid disks, the pitch circle would be the theoretical circle that meshed gears roll on.

**pitch diameter** The diameter of the pitch circle.

**power** A property that describes how fast energy is being used; in other words, power is energy per unit time.

**sliding friction** The frictional drag force on two dry sliding objects.

**spur gear** A type of circular gear with radial teeth (the most common type of gear).

**static friction** The friction force that must be overcome to get an object at rest to move; for a particular object, static friction is greater than sliding friction.

**torque** A twisting force such as would come from a motor; torque is used in rotational systems just as force is used in linear systems.

**underdamped** A system that has relatively little damping so that it responds quickly and tends to overshoot.

**viscous friction** A drag force experienced when a lubricant is used between the objects so that the objects do not actually touch; the drag force, which is proportional to velocity, comes from the layers of lubricant slipping over each other.

**weight** Technically, a downward force exerted by a mass, caused by gravity.

**worm** The spiral-looking gear in a worm gearbox.

**worm gear** The circular gear that the worm meshes with in a worm gearbox.

## EXERCISES

### Section 5.1

1. What is the difference between *static*, *sliding*, and *viscous friction*?
2. An aluminum casting that weighs 10 lb is resting on a steel surface.
  - a. How much force will be necessary to get it started sliding?
  - b. How much force will be necessary to keep it sliding once it is moving?
3. A oak table that weighs 40 lb is resting on a smooth oak floor.
  - a. How much force will be necessary to get it started sliding?
  - b. How much force will be necessary to keep it sliding once it is moving?
4. A coil spring has a spring constant of 5 lb/in. and a rest length of 3 in. In a machine, the spring is stretched to 4.5 in. and is used to exert a constant force on a lever. How much force does the spring exert?
5. The keys of a certain keyboard depress  $\frac{1}{4}$  in. under 1 oz of force. What is the spring constant?
6. A robot arm must be able to extend 24 in. horizontally in 2 s while carrying a load of 10 lb. How much force is required, assuming the arm is accelerating all the way?
7. Rework Exercise 6 with the more realistic assumption that the arm accelerates for the first 12 in., and then decelerates the last 12 in. The total time should still be 2 s.
8. A robot arm must be able to extend 50 cm horizontally in 2 s while carrying a load of 5 kg. How much force is required, assuming the arm is accelerating all the way?
9. Rework Exercise 8 with the more realistic assumption that the arm accelerates for the first 25 cm and then decelerates the last 25 cm. The total time should still be 2 s.
10. A linear actuator can provide a force of 2 lb. How long will it take to accelerate a 6-lb load to a speed of 5 ft/s? (Assume no friction.)
11. Two rotating parts, one heavy and one light, need to be spun up to the same speed in the same period of time. If they end up at the same speed, why does the heavy part require a bigger motor?
12. A torque of 10 in. · lb is applied to a rotating part. The part has a moment of inertia of  $0.5 \text{ lb} \cdot \text{s}^2 \cdot \text{in}$ . How much time will it take to rotate the part  $180^\circ$ ?
13. A wheel is attached to the end of an electric motor shaft. The wheel has a moment of inertia of  $1.5 \text{ lb} \cdot \text{s}^2 \cdot \text{in}$ . If the motor produces a constant torque of 2 in. · lb, how long will it take the motor to spin the wheel up to 100 rpm?

14. A torque of  $18 \text{ N} \cdot \text{m}$  is applied to a rotating part. The part has a moment of inertia of  $0.078 \text{ kg} \cdot \text{m}^2$ . How much time will it take to rotate the part  $180^\circ$ ?
15. A wheel is attached to the end of an electric motor shaft. The wheel has a moment of inertia of  $0.23 \text{ kg} \cdot \text{m}^2$ . If the motor produces a constant torque of  $0.035 \text{ N} \cdot \text{m}$ , how long will it take the motor to spin the wheel up to 100 rpm?
16. A new disk drive must rotate at 300 rpm. How much torque must the motor supply to go from 0 to 300 rpm in 3 s? The moment of inertia of the drive is  $0.04 \text{ lb} \cdot \text{s}^2 \cdot \text{in}$ .
17. A lever like that shown in Figure 5.13(b) is used to press a cap on a container. To press on the cap, 10 lb of force is required at the handle [ $F_1$  in Figure 5.13(b)] with a stroke of 2 in. The total length of the lever is 18 in., and the distance from  $F_2$  to the pivot is 3 in. Find the force and stroke that results at  $F_2$ .
18. A lever like that shown in Figure 5.13(c) is used as a hand lever brake on a machine. To activate the brake, 5 lb of force is required [ $F_1$  in Figure 5.13(c)] with a stroke of 1.5 in. The total length of the lever is 36 in., and the distance from  $F_2$  to the pivot is 6 in. Find the force and stroke that results at  $F_2$ .

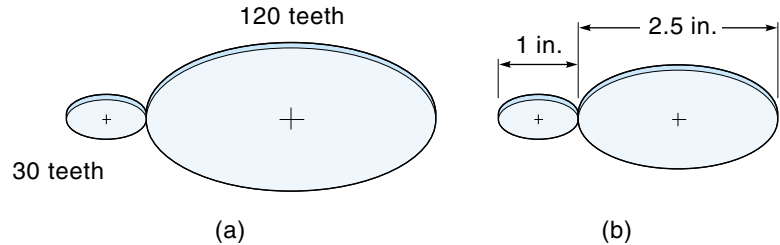
### Section 5.2

19. An electric heater is rated at 220 Vac at 15 A. How many Btu/h of heat does it put out?
20. A small gasoline engine can put out 12,000 ft · lb/min of mechanical power. The engine is used to drive an electric generator that is 80% efficient. Find the output of the generator in watts.
21. A 120-V electric motor is needed that can do the equivalent of lifting a 100-lb weight 10 ft in 1 min. Assuming the motor is 85% efficient, find the motor current.
22. A small gasoline engine can put out 18,000 N · m/min of mechanical power. The engine is used to drive an electric generator that is 80% efficient. Find the output of the generator in watts.
23. A 120-V electric motor is needed that can do the equivalent of lifting a 50-kg mass 3 m in 1 min. Assuming the motor is 85% efficient, find the motor current.
24. An electronic package is to be mounted 6 in. above a diesel engine on a steel bracket. The bracket is  $\frac{1}{4}$  in. thick and 2 in. wide. Assume the engine block is  $175^\circ\text{F}$  and the package is at  $100^\circ\text{F}$ . How much heat (Btu/h) will come through the bracket (neglect heat dissipated by the bracket itself)?

### Section 5.3

25. The shock absorbers on a car dampen out bouncing when the car goes over a bump. Use this example to explain the concept of underdamped, critically damped, and overdamped systems. Specifically, how would the car react to a bump in each case?
26. What may happen if the vibrations in a system—say, from a rotating motor—are at the same frequency as the resonant frequency of one of the system parts?
27. A 2-lb control box is connected by a bracket to a machine. If you push on the control box with 20 lb of force, the bracket bends a little, and the box moves about 0.25 in. Find the resonant frequency of the control box.

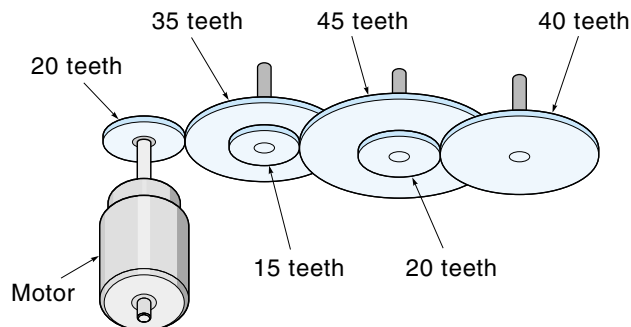
**Figure 5.48**  
(Problems 29–32)



### Section 5.4

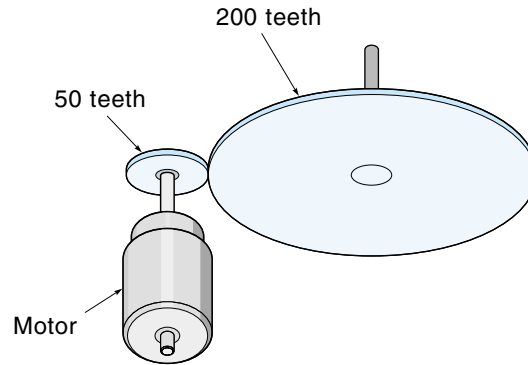
28. What is *diametral pitch*, and why is it an important parameter in a gear mesh?
29. Find the gear ratios of the gears shown in Figure 5.48.
30. For each gear pass shown in Figure 5.48, how many degrees must the small gear turn for the big gear to turn  $90^\circ$ ?
31. For each gear pass shown in Figure 5.48, how many degrees must the big gear turn for the small gear to make two complete revolutions?
32. For each gear pass shown in Figure 5.48, the pinion is rotating at 50 rpm. Find the rpm of the big gear.
33. Find the overall gear ratio of the gear train shown in Figure 5.49. If the motor is rotating at 500 rpm, what is the rotational velocity of the output gear?
34. The motor in Figure 5.50 is turning at 600 rpm with a torque of 2 in. · lb. What is the velocity and torque available at the output shaft?
35. A roller needs to turn at 60 rpm and requires 10 in. · lb of torque. A motor is available with a maximum velocity of 500 rpm at a torque of 1.3 in. · lb. Can the motor be used?
36. How is the moment of inertia of the load affected by a gear pass?
37. Explain the term *backlash*. What is the effect on backlash from long gear trains?
38. Give an advantage and a disadvantage of worm gear drives.

**Figure 5.49**  
(Problem 33)



**Figure 5.50**

(Problem 34)



39. What does the term *lockup* refer to in worm gear drives?
40. What are the advantages of the harmonic drive?
41. What is the gear ratio of a harmonic drive with 200 teeth in the outer cup and 196 teeth on the flex spline?

### Section 5.5

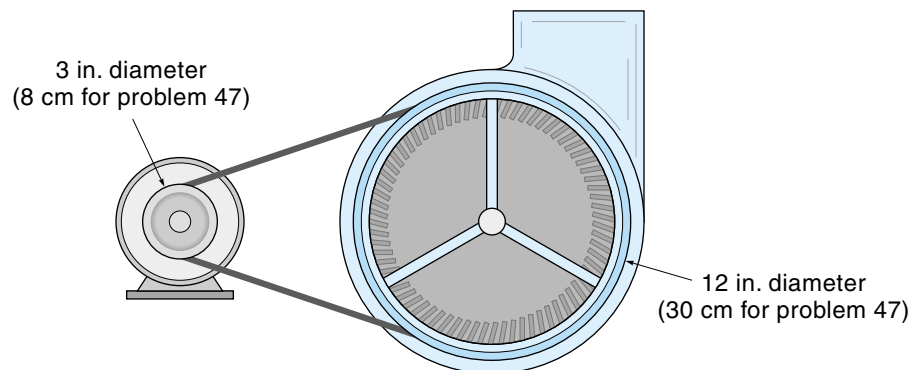
42. Describe the function and operation of an electromagnetic clutch.
43. Describe the function and operation of an electromagnetic brake.

### Section 5.6

44. What are some advantages of transmitting power by belts?
45. A “squirrel-cage” fan is driven by belt from a motor that turns at 1750 rpm (Figure 5.51). The motor pulley is 3 in. in diameter, and the fan pulley is 12 in. At what speed does the fan rotate?

**Figure 5.51**

(Problems 45, 47)



46. The pulleys in a variable-speed transmission expand from 4 to 8 in. in diameter. Find the range of the velocity ratio.
47. A “squirrel-cage” fan is driven by belt from a motor that turns at 1750 rpm (Figure 5.51). The motor pulley is 8 cm in diameter, and the fan pulley is 30 cm. At what speed does the fan rotate?
48. The pulleys in a variable-speed transmission expand from 6 to 20 cm in diameter. Find the range of the velocity ratio.
49. What are some similarities and differences between belts and roller chains as used in power transmission?

# CHAPTER 6

## Sensors

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### OBJECTIVES

After studying this chapter, you should be familiar with the characteristics and operation of such sensors as:

- Position sensors including potentiometers, optical rotary encoders, and linear variable differential transformers.
- Velocity sensors including optical and direct current tachometers.
- Proximity sensors including limit switches, optical proximity switches, and Hall-effect switches.
- Load sensors including bonded-wire strain gauges, semiconductor force strain gauges, and low-force sensors.
- Pressure sensors including Bourdon tubes, bellows, and semiconductor pressure sensors.
- Temperature sensors including bimetallic temperature sensors, thermocouples, resistance temperature detectors, thermistors, and IC temperature sensors.
- Flow sensors including orifice plates, venturis, pitot tubes, turbines, and magnetic flowmeters.
- Liquid-level sensors including discrete and continuous types.

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### INTRODUCTION

The devices that inform the control system about what is actually occurring are called **sensors** (also known as transducers). As an example, the human body has an amazing sensor system that continually presents our brain with a reasonably complete picture of the environment—whether we need it all or not. For a control system, the designer must ascertain exactly what parameters need to be monitored—for example, position, temperature, and pressure—and then specify the sensors and data interface circuitry to do the job. Many times a choice is possible. For example, we might measure fluid flow in a pipe with a flowmeter, or we could measure the flow indirectly by seeing how long

it takes for the fluid to fill a known-sized container. The choice would be dictated by system requirements, cost, and reliability.

Most sensors work by converting some physical parameter such as temperature or position into an electrical signal. This is why sensors are also called **transducers**, which are devices that convert energy from one form to another.

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## 6.1 POSITION SENSORS

Position sensors report the physical position of an object with respect to a reference point. The information can be an angle, as in how many degrees a radar dish has turned, or linear, as in how many inches a robot arm has extended.

### Potentiometers

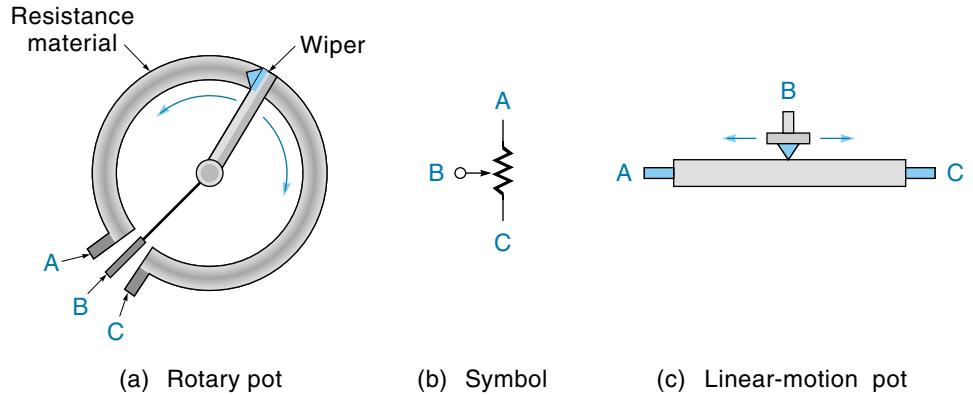
A **potentiometer** (pot) can be used to convert rotary or linear displacement to a voltage. Actually, the pot itself gives resistance, but as we will see, this resistance value can easily be converted to a voltage. Pots used for position sensors are the same in principle as a standard “volume-control,” but there is a difference. A pot used for a volume control may have *audio taper*, which means the resistance changes in a nonlinear fashion to match the human perception of “getting louder.” A pot used to measure angular position has *linear taper*, which means the resistance changes linearly with shaft rotation.

Figure 6.1(a) illustrates how the pot works. A resistive material, such as conductive plastic, is formed in the shape of a circle (terminating at contacts A and C). This material has a very uniform resistivity so that the ohms-per-inch value along its length is a constant. Connected to the shaft is the **slider**, or **wiper**, which slides along the resistor and taps off a value [contact B in Figure 6.1(a)]. Figure 6.1(b) shows the circuit symbol. The pot just described is the single-turn type, which actually has only about 350° of useful range. A single-turn pot may have “stops” at each end of its travel. Obviously, such a pot could only be used where the rotation never exceeds 350°. A single-turn pot without stops has a small “dead zone” when the wiper crosses the end of the resistor. Multiturn pots are available with a wiper that moves in a helix motion, allowing for up to 25 or more revolutions of the shaft from stop to stop. Figure 6.1(c) illustrates a linear-motion potentiometer. In this case, the wiper can move back-and-forth in a straight line. Linear-motion pots are useful for sensing the position of objects that move in a linear fashion.

Figure 6.2(a) shows a pot that detects the angular position of a robot arm. In this case, the pot body is held stationary, and the pot shaft is connected directly to the motor shaft. Ten volts is maintained across the (outside) terminals of the pot. Look at Figure 6.2(b) and imagine how the voltage is changing evenly from 0 to 10 Vdc along the resistive element. The wiper merely taps off the voltage drop between its contact point and ground. For example, if the wiper is at the bottom, the output is 0 V corresponding to



**Figure 6.1**  
Potentiometer.



0°. When the wiper is at the top, the output is 10 V corresponding to 350°; in the exact middle, a 5-V output indicates 175° ( $350^\circ/2 = 175^\circ$ ). Example 6.1 demonstrates how to calculate the pot voltage for any particular angle.

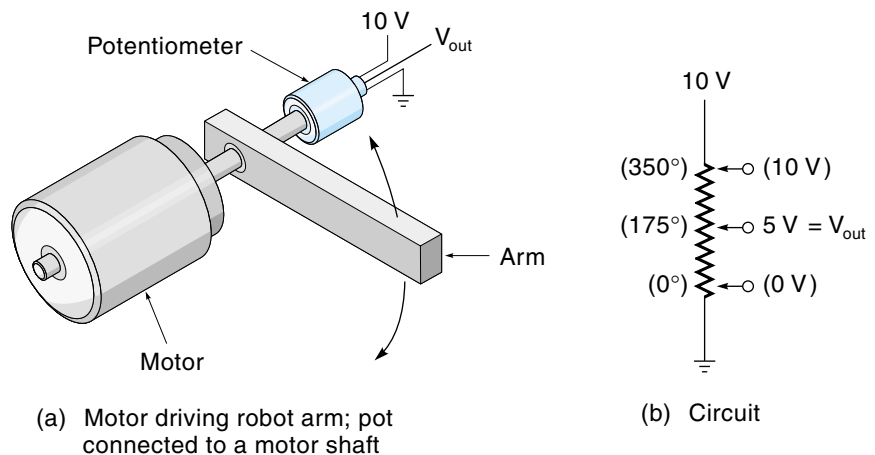
**EXAMPLE 6.1**

A pot is supplied with 10 V and is set at 82° [similar to Figure 6.2(b)]. The range of this single-turn pot is 350°. Calculate the output voltage.

**SOLUTION**

If the pot is supplied 10 V, then the maximum angle of 350° will produce a 10-V output. Using these values, we can set up a ratio of output to input and use that ratio to calculate the output for any input [this ratio is an example of a simple transfer function (TF) discussed in Chapter 1]:

**Figure 6.2**  
Potentiometer as a position sensor.



$$TF_{\text{pot}} = \frac{\text{output}}{\text{input}} = \frac{10 \text{ Vdc}}{350^\circ}$$

To find the output voltage for a particular angle, multiply the angle with the transfer function (and as always, be sure the units work out correctly—in this case, *degrees* cancel, leaving *volts* as the unit):

$$\text{Pot voltage (at } 82^\circ) = \frac{10 \text{ Vdc}}{350^\circ} \times 82^\circ = 2.34 \text{ Vdc}$$

The potentiometer circuit being discussed here is actually a voltage divider, and to work properly the same current must flow through the entire pot resistance. A **loading error** occurs when the pot wiper is connected to a circuit with an input resistance that is *not* considerably higher than the pot's resistance. When this happens, current flows out through the wiper arm, robbing current from the lower portion of the resistor and causing the reading to be low (see Example 6.2). To solve this problem, a high-impedance buffer circuit such as the voltage follower (discussed in Chapter 3) can be inserted between the pot and the circuit it must drive. Loading error is the difference between the unloaded and loaded output as given in Equation 6.1a:

$$\text{Loading error} = V_{\text{NL}} - V_{\text{L}} \quad (6.1a)$$

where

$V_{\text{NL}}$  = output voltage with no load

$V_{\text{L}}$  = output voltage with load applied

### EXAMPLE 6.2

A 10-k $\Omega$  pot is used as a position sensor (Figure 6.3). Assume that the wiper is in the middle of its range. Find the loading error when

- The interface circuit presents an infinite resistance.
- The interface circuit presents a resistance of 100 k $\Omega$ .

### SOLUTION

- Figure 6.3(a) shows the ideal situation where the interface circuit resistance is so high that there is virtually no current in the pot wiper wire. The pot will behave like two 5-k $\Omega$  resistors in series, and we can use the voltage-divider rule to calculate the pot voltage:

$$V_{\text{pot}} = 10 \text{ V} \times \frac{5 \text{ k}\Omega}{5 \text{ k}\Omega + 5 \text{ k}\Omega} = 5 \text{ V}$$

As we would expect, the pot voltage is exactly half of the 10-V supply voltage. There is no loading error in this case.

- b. Now consider the case where the input resistance of the interface circuit is 100 k $\Omega$ , as shown in Figure 6.3(b). We will use the voltage-divider rule again to compute the pot voltage, but this time the lower resistance is the parallel combination of 5 k $\Omega$  and 100 k $\Omega$  (as shown in [Figure 6.3(c)]):

$$5 \text{ k}\Omega // 100 \text{ k}\Omega = \frac{1}{\frac{1}{5 \text{ k}\Omega} + \frac{1}{100 \text{ k}\Omega}} = 4.76 \text{ k}\Omega$$

which is the equivalent lower resistance. Using this value in the voltage divider, we now recalculate the pot voltage:

$$V_{\text{pot}} = 10 \text{ V} \times \frac{4.76 \text{ k}\Omega}{5 \text{ k}\Omega + 4.76 \text{ k}\Omega} = 4.88 \text{ V}$$

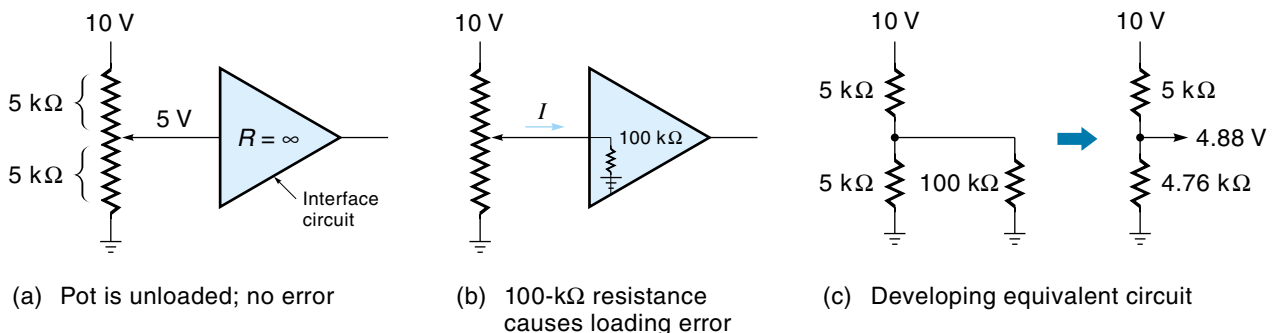
Thus, the actual pot voltage is only 4.88 V when it should be 5 V. The loading error is

$$5 \text{ V} - 4.88 \text{ V} = .12 \text{ V}$$

The maximum loading error occurs when the pot is  $\frac{2}{3}$  of full range. If you were to rework this problem for a pot voltage of 2.5 V, you would find the error is only 0.045 V. Therefore, the effect of loading errors is not linear.

**Figure 6.3**

Loading errors.



In many applications, the total rotary movement to be measured is less than a full revolution. Consider the arm in Figure 6.4 that moves through an angle of only  $90^\circ$ . Using as much of the pot's range as possible in order to get a lower average error rate is advantageous, so we might use a 3 : 1 gear ratio that causes the pot to turn through  $270^\circ$ . (In Figure 6.4, the small pot gear must make three revolutions for each revolution of the motor gear.) The controller will be programmed to understand that  $3^\circ$  of the pot corresponds to only  $1^\circ$  of the actual arm.

As in all physical systems, we must be aware of certain errors that creep in. In this case, carbon pots cannot be made perfectly linear, so we define **linearity error** as the difference between what the angle really is and what the pot reports it to be. The graph of Figure 6.5 shows the ideal versus actual resistance ( $R$ ) for a pot position sensor. The error is the difference in resistance between these two lines. Notice that the error is not the same everywhere, but the maximum error is designated as  $\Delta R$ . Linearity error is defined in percentage, as shown below, and ranges between 1.0 and 0.1% (but higher precision costs more, of course):

$$\text{Linearity error} = \frac{\Delta R \times 100}{R_{\text{tot}}} \quad (6.1b)$$

where

$\Delta R$  = maximum resistance error

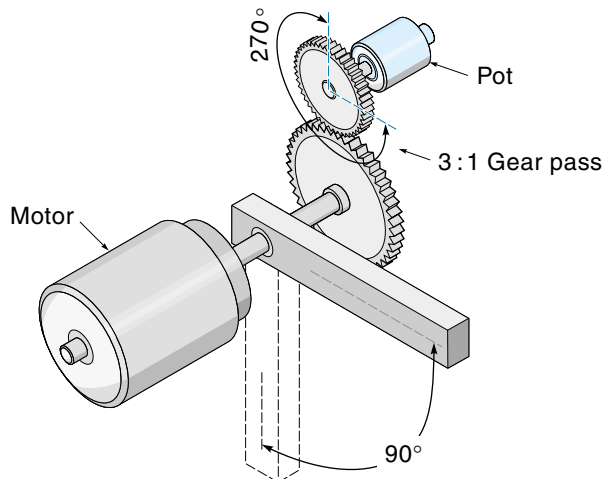
$R_{\text{tot}}$  = total pot resistance

When the potentiometer is used as a position sensor, the output voltage is directly proportional to the shaft angular position, so linearity error can also be expressed in terms of angle:

$$\text{Linearity error} = \frac{\Delta\theta \times 100}{\theta_{\text{tot}}} \quad (6.1c)$$

#### Figure 6.4

When motor shaft is restricted to  $90^\circ$ , the 3:1 gear pass turns the pot through  $270^\circ$ .



where

$\Delta\theta$ = maximum angle error (in degrees)

$\theta_{\text{tot}}$ = total range of the pot (in degrees)

(Note: Loading effects will also contribute to the error.)

### EXAMPLE 6.3

A single-turn pot (350°) has a linearity error of 0.1% and is connected to a 5 Vdc source. Calculate the maximum angle error that could be expected from this system.

### SOLUTION

To calculate the maximum possible angle error, rearrange Equation 6.1c and solve:

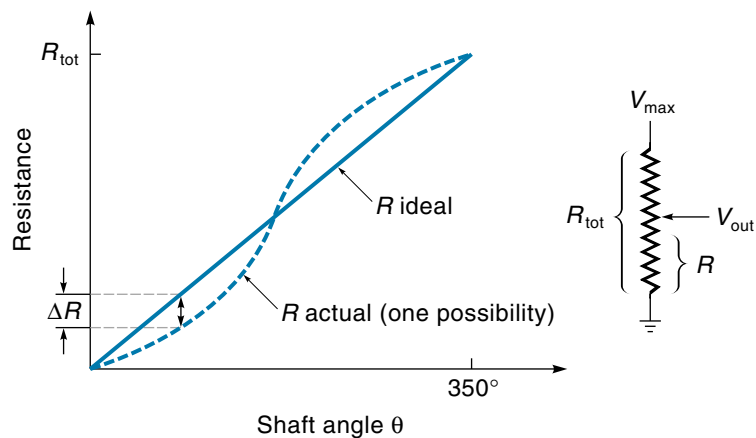
$$\Delta\theta = \frac{\text{linearity} \times \theta_{\text{tot}}}{100} = \frac{0.1 \times 350^\circ}{100} = 0.35^\circ$$

If this pot were in a control system, the controller would only know the position to within 0.35° or about one-third of a degree.

Linearity error determines the accuracy of a sensor. A related but different measurement concept is resolution. **Resolution** refers to the smallest increment of data that can be detected and/or reported. In digital systems, the resolution usually refers to the value of the **least significant bit** (LSB) because that is the smallest change that can be reported. For example, a 2-bit number has four possible states (00, 01, 10, 11). If we used this 2-bit number to quantify the gas level in your tank, we could specify the

**Figure 6.5**

Linearity error of a pot: ideal vs. actual.



amount of gas only to the nearest fourth, that is, empty, one-quarter, one-half, and three-quarters. Thus, the resolution would be one-fourth of a tank of gas. The “accuracy” of a digital system should be  $\pm\frac{1}{2}$  LSB (although it could be worse, in which case the LSB wouldn’t mean very much). In the gas-gauge example,  $\pm\frac{1}{2}$  LSB accuracy corresponds to  $\pm\frac{1}{8}$  tank, so if the gauge reads half full, you would know that the actual level was between three-eighths and five-eighths full.

For an analog device such as a potentiometer, resolution refers to the smallest change that can be measured. It is usually expressed in percentage:

$$\begin{aligned}\% \text{ resolution} &= \frac{\text{smallest change in resistance} \times 100}{\text{total resistance}} \\ &= \frac{\Delta R}{R_{\text{tot}}} \times 100\end{aligned}\quad (6.2)$$

Let’s examine resolution in conjunction with the **wire-wound potentiometer**. A wire-wound pot uses a coil of resistance wire for the resistive element (see Figure 6.6). The wiper bumps along on the top of the coil. Clearly, the resolution in this case is the resistance of one loop of the coil. This concept is illustrated in Example 6.4.

#### EXAMPLE 6.4

The resistive element of a wire-wound pot is made from 10 in. of 100  $\Omega$ /in. resistance wire and is wound as a coil of 200 loops. The range of the pot is 350°. What is the resolution of this pot?

#### SOLUTION

$$\text{Total resistance} = R_{\text{tot}} = 10 \text{ in.} \times \frac{100 \Omega}{\text{in.}} = 1 \text{ k}\Omega$$

The pot coil has 200 turns of wire. Therefore, the smallest increment of resistance corresponds to one loop of the coil. The resistance of one loop of wire is

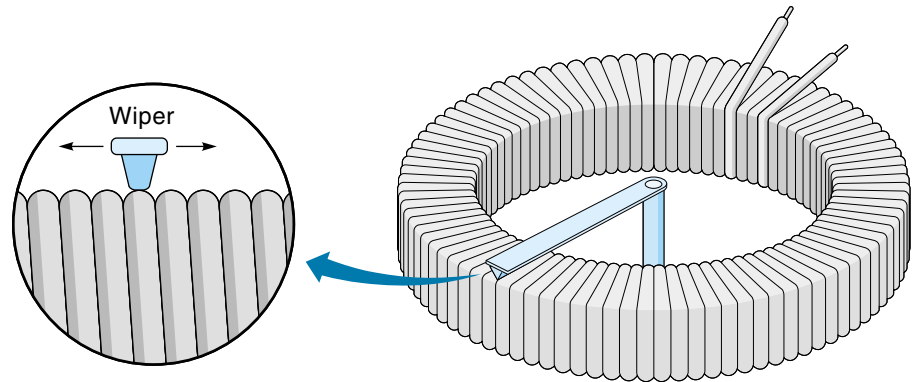
$$\text{Resistance/loop} = \frac{1 \text{ k}\Omega}{200 \text{ loops}} = 5 \Omega/\text{loop}$$

Thus, resolution is

$$\frac{\Delta R \times 100}{R_{\text{tot}}} = \frac{5 \Omega \times 100}{1 \text{ k}\Omega} = 0.5\%$$

If this pot were to be used as a position sensor, it would be useful to know what the resolution is in degrees. The smallest measurable change corresponds to one loop of the resistance coil, and this pot divides 350° into 200 parts; therefore, the resolution in degrees would be 350°/200 loops = 1.75°.

**Figure 6.6**  
Resolution in a  
wire-wound pot.



The output of a position sensor should be a continuous DC voltage, but the slider action of pots can sometimes cause voltage transients. This is particularly true for wire-wound pots because the slider may momentarily break contact as it bumps from wire to wire. If this is a problem, it can usually be resolved with a low-pass filter, which is simply a capacitor to ground (Figure 6.7). The capacitor stays charged up to the average pot voltage and resists momentary voltage changes.

Example 6.5 uses a potentiometer as the position sensor for a digital feedback control system. The main consideration here is resolution from the analog to the digital.

### EXAMPLE 6.5

The robot arm illustrated in Figure 6.7 rotates  $120^\circ$  stop-to-stop and uses a pot as the position sensor. The controller is an 8-bit digital system and needs to know the actual position of the arm to within  $0.5^\circ$ . Determine if the setup shown in Figure 6.7 will do the job.

### SOLUTION

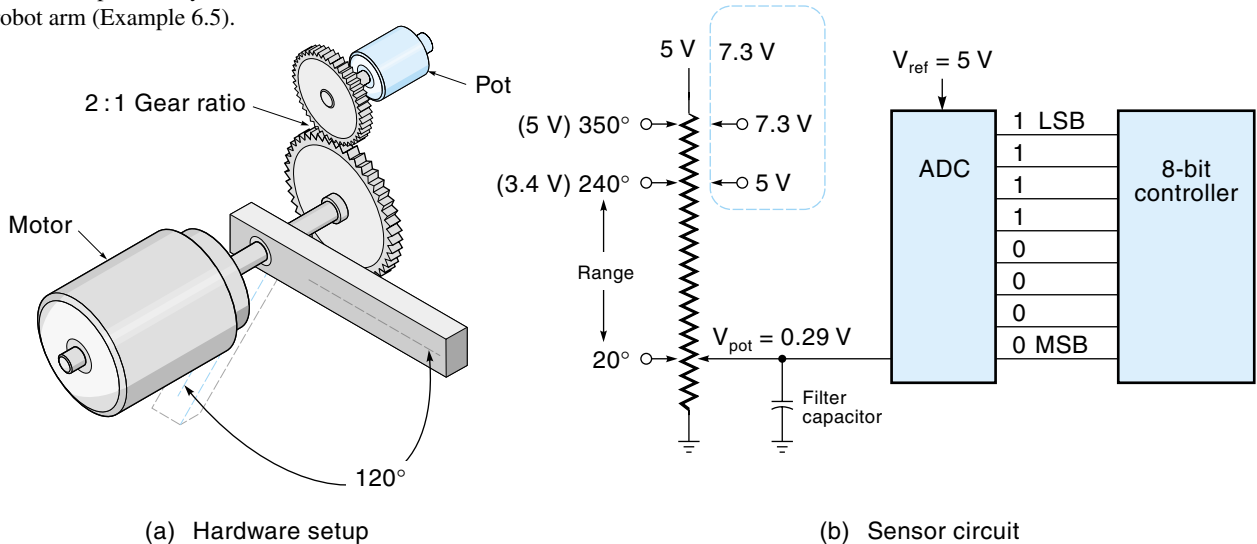
To have  $0.5^\circ$  resolution means that the entire  $120^\circ$  will be divided into 240 increments, each increment being  $0.5^\circ$ . An 8-bit number has 255 levels (from 00000000 to 11111111), so it has more than enough to do the job. (That's good!)

The pot is supplied with 5 V. Therefore, the output of the pot would be 5 V for the maximum pot angle of  $350^\circ$  (if it could rotate that far). Notice that the reference voltage of the ADC (analog-to-digital converter) is also set at 5 V; thus, if the pot voltage ( $V_{\text{pot}}$ ) is 5 V, the digital output would be 255 ( $11111111_{\text{bin}}$ ). Table 6.1 summarizes this (see last three columns).

A single-turn pot has a range of  $350^\circ$ , but the robot arm only rotates  $120^\circ$ , hence the 2 : 1 gear ratio between the pot and the arm. With this arrangement, the pot rotates  $240^\circ$  when the arm rotates  $120^\circ$ . By doubling the operating

**Figure 6.7**

Pot sensor position system for robot arm (Example 6.5).



range of the pot, the linearity and resolution errors (from the pot) are reduced by half.

Consider the case when the robot arm is at  $10^\circ$  (second line in Table 6.1). Because of the 2 : 1 gear ratio, the pot would be at  $20^\circ$ . To calculate the pot voltage at  $20^\circ$ , we use the transfer function of the pot ( $5\text{ V}/350^\circ$ ):

$$V_{\text{pot}} = \frac{5\text{ V}}{350^\circ} \times 20^\circ = 0.29\text{ V}$$

This  $0.29\text{ V}$  is then converted into binary with the ADC (see Figure 6.7). To calculate the binary output, first form the ADC transfer function:

$$\frac{\text{output}}{\text{input}} = \frac{255\text{ states}}{5\text{ V}}$$

Now calculate the ADC binary output using the  $0.29\text{ V}$  (pot voltage) as the input:

$$\frac{255\text{ states}}{5\text{ V}} \times 0.29\text{ V} = 14.8 \approx 15_{\text{states}} = 00001111_{\text{bin}}$$

We now turn our attention to the system resolution, which is the smallest measurable change. In a digital system, this usually corresponds to the value assigned to the LSB (you can't change half a bit!). We can find the resolution by calculating the pot angle corresponding to a single binary state. This is done by multiplying the transfer functions of each of the system elements together



to get an overall system transfer function (you may notice that we actually used the inverse of the transfer functions in order to get the desired units):

$$\underbrace{\frac{1^\circ}{2^\circ}}_{\text{Gears}} \times \underbrace{\frac{350^\circ}{5 \text{ V}}}_{\text{Pot}} \times \underbrace{\frac{5 \text{ V}}{255 \text{ states}}}_{\text{ADC}} = \frac{0.686^\circ}{\text{state}}_{\text{arm}}$$

This result tells us that the LSB of the ADC is  $0.686^\circ$ , which is too big! We need the LSB to be  $0.5^\circ$ . As it stands, this design does not meet the specification. Can it be fixed? Yes, looking back, you can see that at  $350^\circ$  the pot sends 5 V to the ADC, but this will *never* happen because the pot is constrained to  $240^\circ$ . To get maximum resolution from the ADC, the pot should send 5 V to the ADC when the pot is  $240^\circ$ . This will require raising the pot supply voltage to 7.3 V [by ratio,  $5 \text{ V} \times (350^\circ/240^\circ) = 7.3 \text{ V}$ ].

The revised voltages are shown in the dashed circle in Figure 6.7. Now the resolution is recalculated to be

$$\frac{1^\circ}{2^\circ} \times \frac{350^\circ}{7.3 \text{ V}} \times \frac{5 \text{ V}}{255 \text{ states}} = 0.470^\circ/\text{state}$$

This result is within the  $0.5^\circ$  specification for resolution.

## Optical Rotary Encoders

An **optical rotary encoder** produces angular position data directly in digital form, eliminating any need for the ADC converter. The concept is illustrated in Figure 6.8, which shows a slotted disk attached to a shaft. A light source and photocell arrangement are mounted so that the slots pass the light beam as the disk rotates. The angle of the shaft is deduced from the output of the photocell. There are two types of optical rotary encoders: the absolute encoder and the incremental encoder.

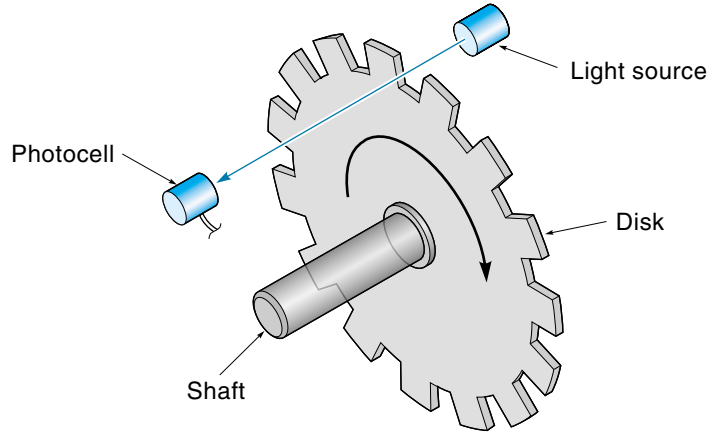
**TABLE 6.1**

**System Values for Various Angles of Robot Arm**

Arm angle (degrees)	Pot angle (degrees)	Pot voltage (V)	ADC output (binary states)
0	0	0	0000000
10	20	0.29	00001111
120	240	3.43	10110000
175	350	5	11111111

**Figure 6.8**

An optical rotary encoder.

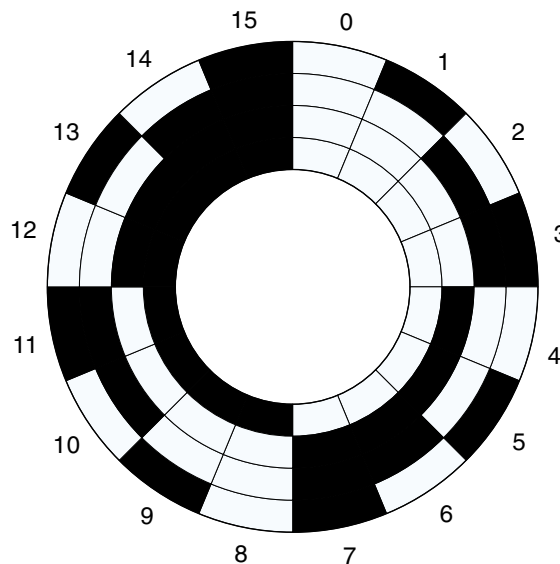


### *Absolute Optical Encoders*

**Absolute optical encoders** use a glass disk marked off with a pattern of concentric tracks (Figure 6.9). A separate light beam is sent through each track to individual photo sensors. Each photo sensor contributes 1 bit to the output digital word. The encoder in Figure 6.9 outputs a 4-bit word with the LSB coming from the outer track. The disk is divided into 16 sectors, so the resolution in this case is  $360^\circ/16 = 22.5^\circ$ . For better resolution, more tracks would be required. For example, eight tracks (providing 256 states) yield  $360^\circ/256 = 1.4^\circ/\text{state}$ , and ten tracks (providing 1024 states) yield  $360^\circ/1024 = 0.35^\circ/\text{state}$ .

**Figure 6.9**

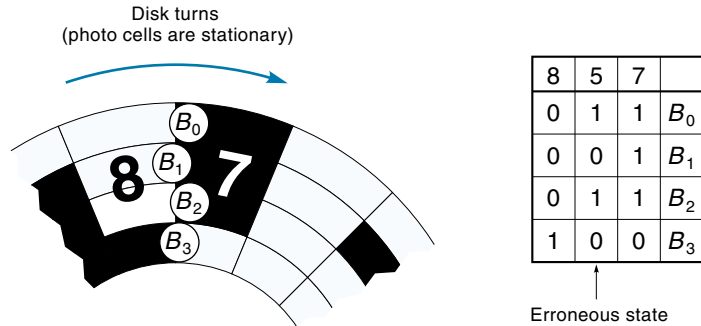
An absolute optical encoder using straight binary code.



(Note: Black areas cause a 1 output)

**Figure 6.10**

An absolute optical encoder showing how an out-of-alignment photocell can cause an erroneous state. (Note: Dark areas produce a 1, and light areas produce a 0.)

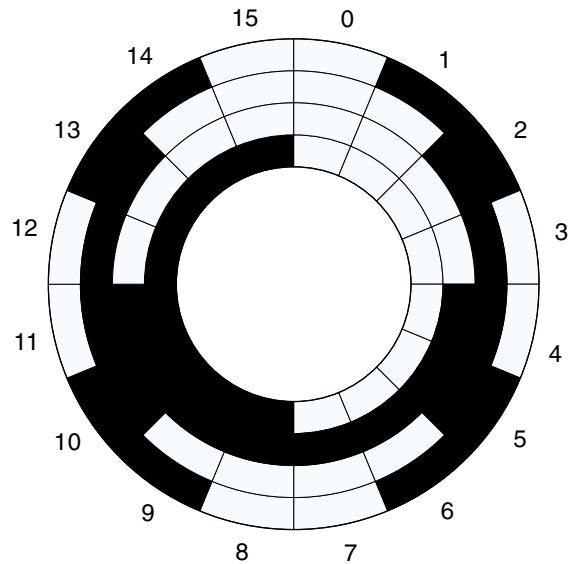


An advantage of this type of encoder is that the output is in straightforward digital form and, like a pot, always gives the absolute position. This is in contrast to the incremental encoder that, as will be shown, provides only a relative position. A disadvantage of the absolute encoder is that it is relatively expensive because it requires that many photocells be mounted and aligned very precisely.

If the absolute optical encoder is not properly aligned, it may occasionally report completely erroneous data. Figure 6.10 illustrates this situation, and it occurs when more than 1 bit changes at a time, say, from sector 7 (0111) to 8 (1000). In the figure, the photo sensors are not exactly in a straight line. In this case, sensor  $B_1$  is out of alignment (it's ahead) and switches from a 1 to a 0 before the others. This causes a momentary erroneous output of 5 (0101). If the computer requests data during this "transition" time, it would get the wrong answer. One solution is to use the **Grey code** on the disk instead of the straight binary code (Figure 6.11). With the Grey code, only 1 bit changes

**Figure 6.11**

An absolute optical encoder using a grey code.



between any two sectors. If the photocells are out of line, the worst that could happen is that the output would switch early or late. Put another way, the error can never be more than the value of 1 LSB when using the Grey code.

### *Incremental Optical Encoders*

The **incremental optical encoder** (Figure 6.12) has only one track of equally spaced slots. Position is determined by counting the number of slots that pass by a photo sensor, where each slot represents a known angle. This system requires an initial reference point, which may come from a second sensor on an inner track or simply from a mechanical stop or limit switch. In many applications, the shaft being monitored will be cycling back-and-forth, stopping at various angles. To keep track of the position, the controller must know which direction the disk is turning as well as the number of slots passed. Example 6.6 illustrates this.

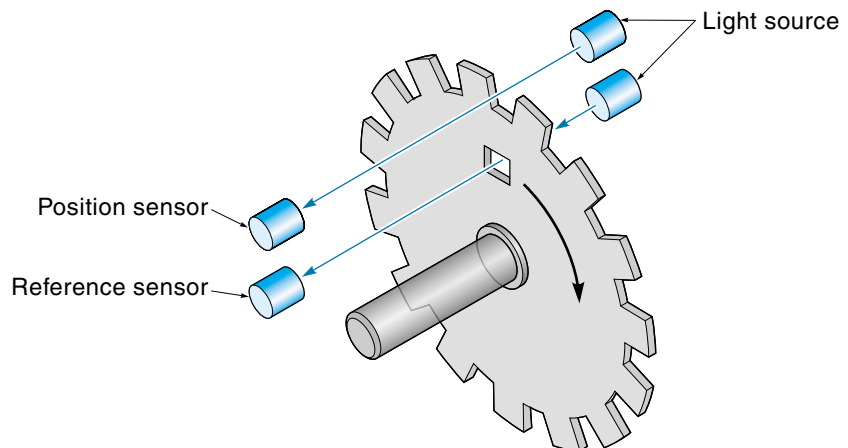
#### **EXAMPLE 6.6**

An incremental encoder has 360 slots. Starting from the reference point, the photo sensor counts 100 slots clockwise (CW), 30 slots counterclockwise (CCW), then 45 slots CW. What is the current position?

#### **SOLUTION**

If the disk has 360 slots, then each slot represents  $1^\circ$  of rotation. Starting at the reference point, we first rotated  $100^\circ$  CW, then reversed  $30^\circ$  to  $70^\circ$ , and finally reversed again for  $45^\circ$ , bringing us finally to  $115^\circ$  (CW) from the reference point.

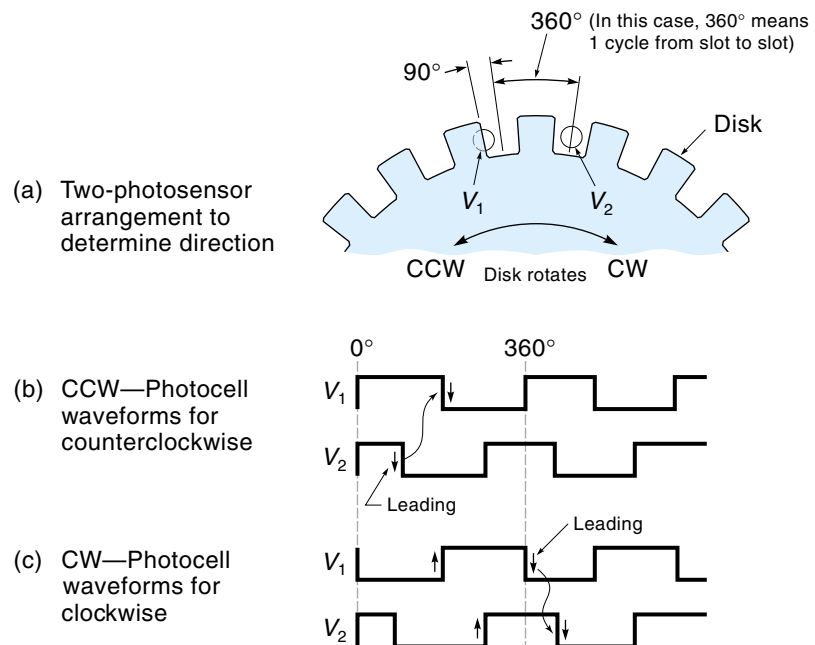
**Figure 6.12**  
An incremental  
optical encoder.



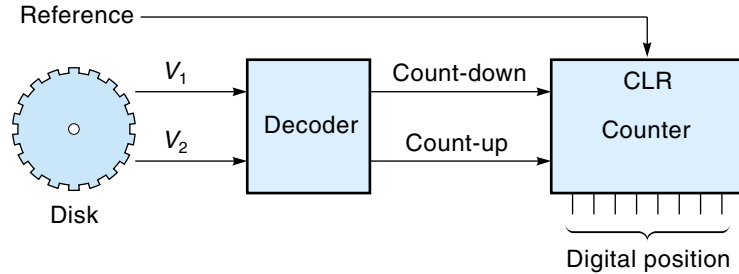
A single photo sensor cannot convey which direction the disk is rotating; however, a clever system using two sensors can. As Figure 6.13(a) illustrates, the two sensors,  $V_1$  and  $V_2$ , are located slightly apart from each other on the same track. For this example,  $V_1$  is initially off (well, almost—you can see it is half-covered up), and  $V_2$  is on. Now imagine that the disk starts to rotate CCW. The first thing that happens is that  $V_1$  comes completely on (while  $V_2$  remains on). After more rotation,  $V_2$  goes off, and slightly later  $V_1$  goes off again. Figure 6.13(b) shows the waveform for  $V_1$  and  $V_2$ . Now consider what happens when the disk is rotated in the CW direction [starting again from the position shown in Figure 6.13(a)]. This time  $V_1$  goes off immediately, and  $V_2$  stays on for half a slot and then goes off. Later  $V_1$  comes on, followed by  $V_2$  coming on. Figure 6.13(c) shows the waveforms generated by  $V_1$  and  $V_2$ . Compare the two sets of waveforms—notice that *in the CCW case  $V_2$  leads  $V_1$  by  $90^\circ$ , whereas for the CW case  $V_1$  is leading  $V_2$  by  $90^\circ$* . This difference in phase determines which direction the disk is turning.

**Decoding  $V_1$  and  $V_2$**  The hardware of the incremental encoder is simpler than for the absolute type. The price paid for that simplicity is that we do not get direct binary position information from  $V_1$  and  $V_2$ . Instead, a decoder circuit must be employed to convert the signals from the photo sensors into a binary word. Actually, the circuit has two parts: The first part extracts direction information, and the second part is an up-down counter, which maintains the slot count. The block diagram of Figure 6.14 shows this. Referring to the diagram, we see that  $V_1$  and  $V_2$  are converted into two new signals

**Figure 6.13**  
An incremental  
optical encoder.



**Figure 6.14**  
Block diagram of an incremental encoder system.

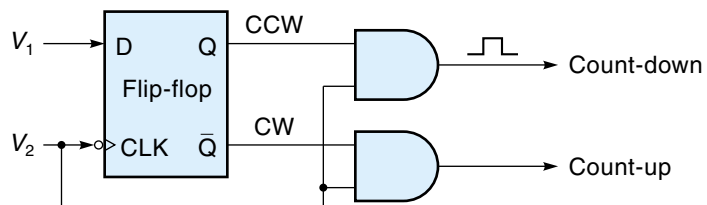


denoted by “count-down” and “count-up”. The count-down signal gives one pulse for every slot passed when the disk is going counterclockwise. The count-up signal gives one pulse for each slot when the disk is rotating clockwise. These signals are then fed to an up-down counter such as the TTL 74193. This counter starts out at 0 (it is usually reset by the reference sensor) and then proceeds to maintain the position by keeping track of the CCW and CW counts. Referring again to Example 6.6, the counter would start at 0, count up to 100, count down 30 pulses to 70, and then count up 45 pulses to 115. Thus, the accumulated total on the counter always represents the current absolute position.

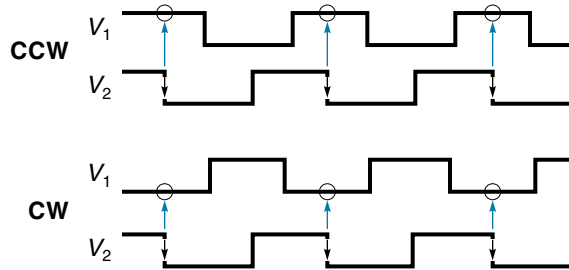
The simplest way to perform the decoding is with a single D-type flip-flop and two AND gates (Figure 6.15). To understand how this circuit functions, we need to examine the waveforms of  $V_1$  and  $V_2$  (Figure 6.16). In the CCW case, every time  $V_2$  goes low,  $V_1$  is high; in the CW case, when  $V_2$  goes low,  $V_1$  is low. This fact is used to separate CCW and CW rotation.  $V_2$  is connected to the negative-going clock of the flip-flop, and  $V_1$  is connected to the D input. Every time  $V_2$  goes low,  $V_1$  is latched and appears at the output. Thus, as long as the disk is rotating CCW, the output will be high; and as long as it rotates CW, the output will be low. These direction signals can be gated with  $V_2$  to produce the required counter inputs count-up and count-down. The count-up signal pulses once per slot when the disk is turning clockwise, and the count-down signal pulses when the disk is turning counterclockwise.

The decoding described so far is the straightforward low-resolution approach. Getting a resolution four times better with more sophisticated decoding is possible because the signals  $V_1$  and  $V_2$  cycle through four distinct states each time a slot passes the sensors. These states can be seen in Figure 6.17. If we were to decode each of these states in Example 6.7, then we would know the angle to the nearest  $0.36^\circ$  ( $1.44^\circ/4 = 0.36^\circ$ ) instead of  $1.44^\circ$ .

**Figure 6.15**  
A decoder for an incremental optical encoder.



**Figure 6.16**  
Decoding direction  
from  $V_1$  and  $V_2$ .



### EXAMPLE 6.7

A position-sensor system (Figure 6.14) uses a 250-slot disk. The current value of the counter is 00100110. What is the angle of the shaft being measured?

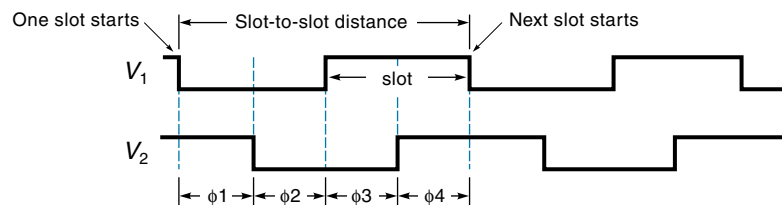
### SOLUTION

For a 250-slot disk, each slot represents  $360^\circ/250 = 1.44^\circ$ , and a count of 00100110 = 38 decimal, so the position is  $38 \times 1.44^\circ = 54.72^\circ$ .

**Interfacing the Incremental Encoder to a Computer** There is a special problem when attempting to pass data to a computer from a standard ripple-type digital counter.\* The counter is counting real-world events and so is not synchronized with the computer. If the computer requests position data while the counter is changing, it may very well get meaningless data. Because this is a remote possibility, the resulting errors are infrequent and in many applications can be ignored. But other situations require that data *always* be accurate.

One approach to the problem might be to disable or “freeze” the counter during the time when the computer is receiving data. But if a count-pulse occurs while the counter

**Figure 6.17**  
An incremental  
encoder showing  
four unique states  
for each slot cycle.



\*The common ripple counter takes a finite time to settle out because a new count may cause a “carry” to ripple up through all bits.

is frozen, it will be lost. The solution is to put a *latch* (a temporary holding register) between the counter and the computer (Figure 6.18). With this setup, the counter is never disabled and always holds the correct count. The latch is connected so that it ordinarily contains the same value as the counter. During those brief times when the counter is counting, the latch is inhibited from changing. With this system, a count is never permanently lost. The worst situation would be if a count came in while a computer exchange was in progress; in this case, the new count would not be reported with the current exchange because the latch is frozen. As soon as the counter finished updating, however, the latch would be updated, and the count would be reported with the next computer exchange.

### EXAMPLE 6.8

The angular position of a shaft must be known to a resolution of  $0.5^\circ$ . A system that uses a 720-slot encoder (Figure 6.19) is proposed. The controller uses a 8051 microcontroller which has 8 bit ports. Will this design meet the specifications?

### SOLUTION

For  $0.5^\circ$  resolution, the encoder must have a slot every  $0.5^\circ$  as a minimum. First, calculate the number of slots required:

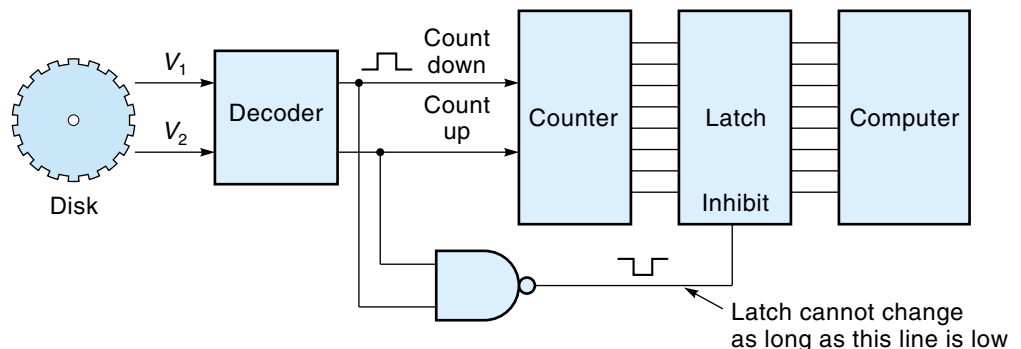
$$\frac{360^\circ}{0.5^\circ/\text{slot}} = 720 \text{ slots}$$

The 720-slot encoder will work just fine. Being a digital system, the resolution is determined by the LSB, which in this case should correspond to 1 slot on the disk ( $0.5^\circ$ ). The binary output should have a range of 0-719 (for 720 states), so the circuit must have the capacity to handle 10-bit data (because it takes 10 bits to express 719).

$$719 \text{ (decimal)} = 1011001111 \text{ (binary)}$$

**Figure 6.18**

An incremental encoder interface circuit showing how the latch is inhibited from changing when the counter is updated.





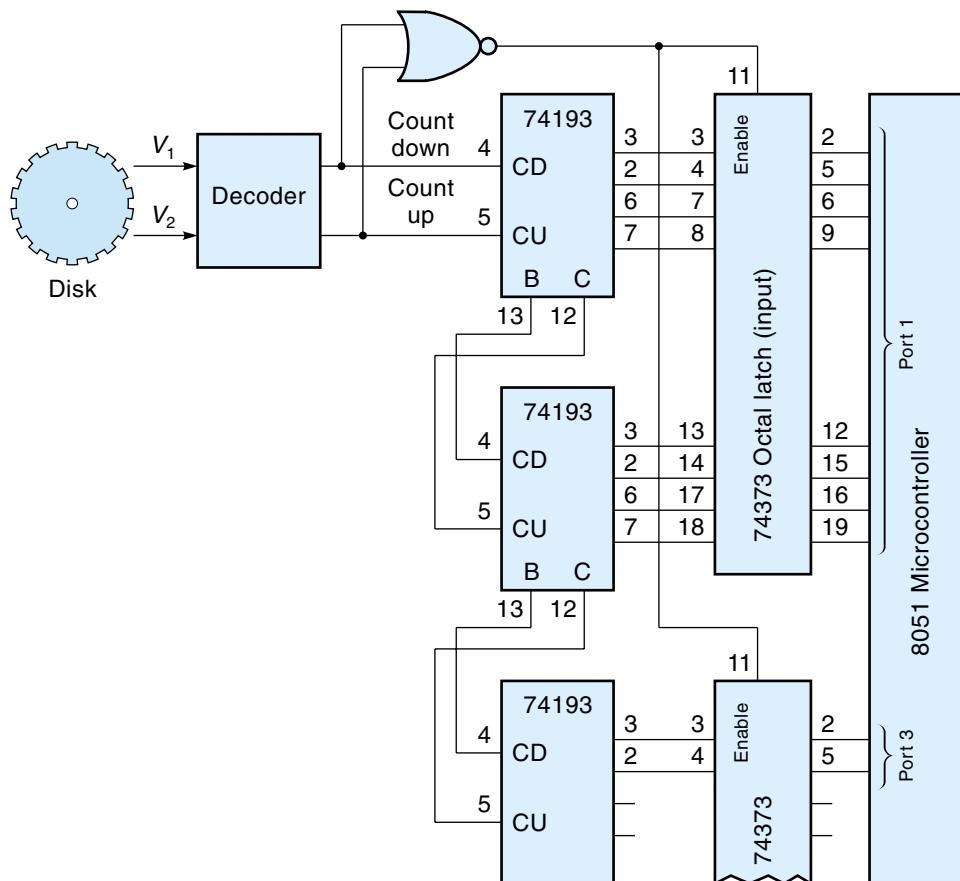
Because the controller is an 8-bit microcontroller, it will require two ports to input the entire 10 bits. As shown in Figure 6.19, the counter consists of three 74193 4-bit up-down counters. The outputs of the counter are constantly updating the 10-bit latch made from two 74373s. The outputs of the latch are connected to ports 1 and 3 of the 8051.

## Linear Variable Differential Transformers

The **linear variable differential transformer (LVDT)** is a high-resolution position sensor that outputs an AC voltage with a magnitude proportional to linear position. It has a relatively short range of about 2 in., but it has the advantage of no sliding contacts. Figure 6.20(a) illustrates that the unit consists of three windings and a movable iron core. The center winding, or *primary*, is connected to an AC reference voltage. The outer two windings, called *secondaries*, are wired to be out of phase with each other and are connected

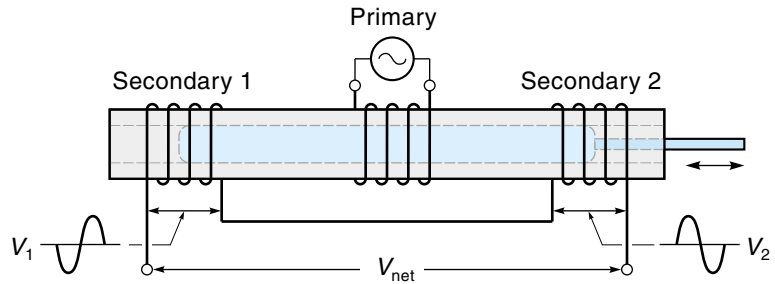
**Figure 6.19**

A circuit diagram of optical-encoder-to-microcontroller interface (Example 6.8).

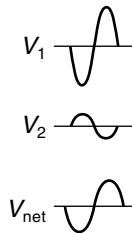


**Figure 6.20**

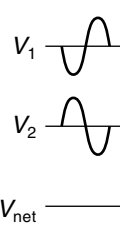
A linear variable differential transformer (LVDT).



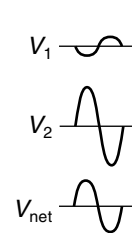
(a) LVDT with shaft centered



(b) Shaft left



(c) Shaft centered



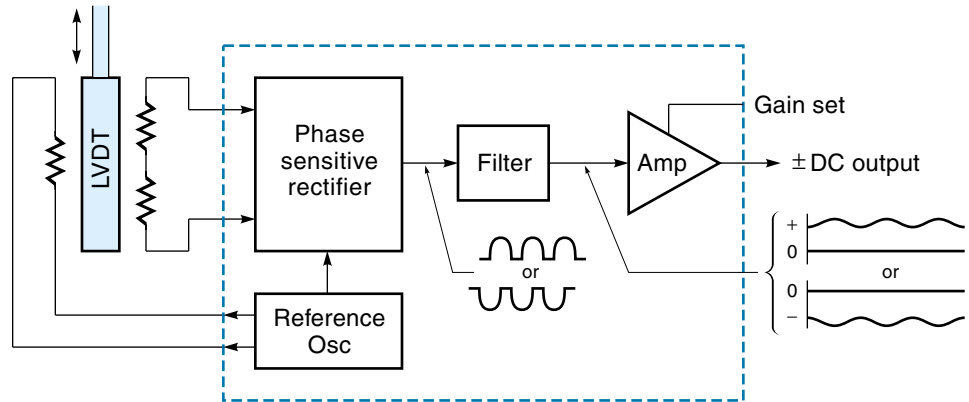
(d) Shaft right

in series. If the iron core is exactly in the center, the voltages induced on the secondaries by the primary will be equal and opposite, giving a net output ( $V_{net}$ ) of 0 V [as shown in Figure 6.20(c)]. Consider what happens when the core is moved a little to the right. Now there is more coupling to secondary 2 so its voltage is higher, while secondary 1 is lower. Figure 6.20(d) illustrates the waveforms of this situation. The algebraic sum of the two secondaries is in phase with secondary 2, and the magnitude is proportional to the distance the core is off center. If the core is moved a little left of center, then secondary 1 has the greater voltage, producing a net output that is in phase with secondary 1 [Figure 6.20(b)]. In fact, the only way we can tell from the output which direction the core moved is by the phase. Summarizing, the output of the LVDT is an AC voltage with a magnitude and phase angle. The magnitude represents the distance that the core is off center, and the phase angle represents the direction of the core (left or right.)

Figure 6.21 illustrates an LVDT with its one-chip support electronics. An oscillator provides the AC reference voltage to the primary—typically, 50-10 KHz at 10 V or less. The output of the LVDT goes first to a phase-sensitive rectifier. This circuit compares the phase of LVDT output with the reference voltage. If they are in phase, the rectifier outputs only the positive part of the signal. If they are out of phase, the rectifier outputs only the negative parts. Next, a low-pass filter smooths out the rectified signal to produce DC. Finally, an amplifier adjusts the gain to the desired level. The output of the LVDT interface circuit is a DC voltage whose magnitude and polarity are proportional to the linear distance that the core is offset from the center. Some integrated

**Figure 6.21**

An interface circuit for an LVDT.



circuits, such as the AD698 (Analog Devices), combine all the functions shown within the box (of Figure 6.21) on a single chip.

## 6.2 ANGULAR VELOCITY SENSORS

*Angular velocity sensors*, or *tachometers*, are devices that give an output proportional to angular velocity. These sensors find wide application in motor-speed control systems. They are also used in position systems to improve their performance.

### Velocity from Position Sensors

Velocity is the rate of change of position. Expressed mathematically,

$$\text{Velocity} = \frac{\Delta\theta}{\Delta t} = \frac{\theta_2 - \theta_1}{t_2 - t_1} \quad (6.3)$$

where

$\Delta\theta$  = change in angle

$\Delta t$  = change in time

$\theta_2, \theta_1$  = position samples

$t_2, t_1$  = times when samples were taken

Because the only components of velocity are position and time, extracting velocity information from two sequential position data samples should be possible (if you know the time between them). This concept is demonstrated in Example 6.9. The math could be done with hard-wired circuits or software. If the system already has a position sensor, such as a potentiometer, using this approach eliminates the need for an additional (velocity) sensor.

**EXAMPLE 6.9**

A rotating machine part has a pot position sensor connected through an ADC such that  $\text{LSB} = 1^\circ$ . Determine how to use this setup to get velocity data.

**SOLUTION**

Velocity can be computed from two sequential position samples— $\theta_1$  taken at time  $t_1$  and  $\theta_2$  taken at  $t_2$ , as specified in Equation 6.3:

$$\text{Velocity} = \frac{\Delta\theta}{\Delta t} = \frac{\theta_2 - \theta_1}{t_2 - t_1}$$

If we took a data sample exactly every second, then the denominator of Equation 6.3 would be 1. In that case, velocity would just equal  $(\theta_2 - \theta_1)$ , but 1 s is probably too long a time for the controller to wait between samples. Instead, select  $\frac{1}{10}$  s (100 ms) as the time between samples. Now,

$$\frac{\Delta\theta}{\Delta t} = \frac{\theta_2 - \theta_1}{1/10} = 10(\theta_2 - \theta_1)$$

Thus, all the software has to do to calculate velocity is

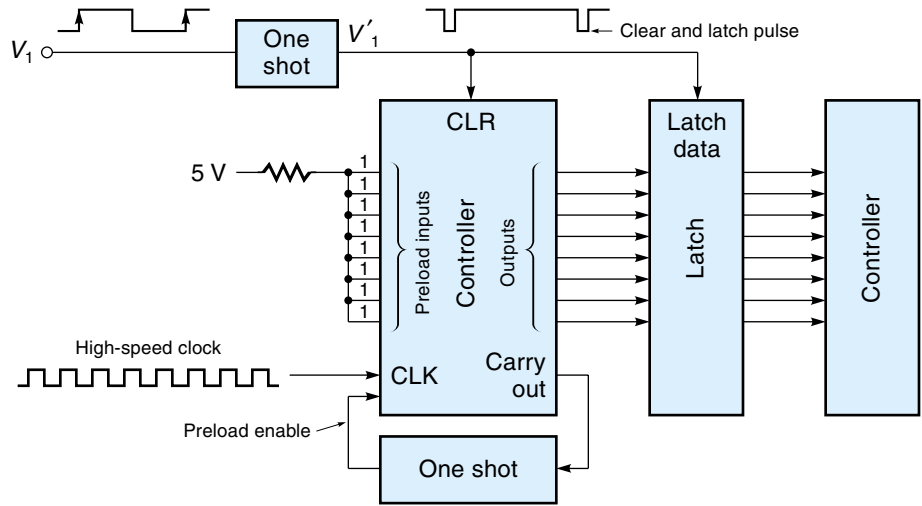
1. Take two position samples exactly  $\frac{1}{10}$  s apart.
2. Subtract the values of the two samples.
3. Multiply the result by 10.

Velocity data can be derived from an optical rotary encoder in two ways. The first would be the method just described for the potentiometer; the second method involves determining the time it takes for each slot in the disk to pass. The slower the velocity, the longer it takes for each slot to go by. The digital counter circuit shown in Figure 6.22 can be used as a timer to time how long it takes for one slot to pass. The idea is to count the cycles of a known high-speed clock for the duration of one *slot period*. The final count would be proportional to the time it took for the slot to pass.

The operation of the circuit (Figure 6.22) is as follows. One of the outputs of the optical encoder (say,  $V_1$ ) is used as the input to the timer.  $V_1$  triggers a one-shot to produce  $V_1'$ , which is a brief negative-going pulse to clear the counter. When  $V_1'$  returns high (removing the clear), a high-speed clock is counted by the counter. When the next slot triggers the one-shot, the counter data are transferred into a separate latch, and the counter is cleared so it can start over again. The controller reads the count from the latch. The value of the count is proportional to the *reciprocal* of the angular velocity. The slower the velocity, the larger the count. This means that for very slow velocities the counter might overflow and start counting up from 0 again (such as your car odometer turning over from 99,999 to 00,000). In fact, when the disk comes to a dead stop, *any* counter would overflow eventually. To solve this problem, a special circuit using another one-shot has been

**Figure 6.22**

Circuit for counting slot-cycle time (for determining velocity from an incremental encoder).



added. Every time the counter fills up, the one-shot fires and reloads 1s into all bits. This action prevents the full counter from ever rolling over to 0. The result is that a full counter is interpreted by the controller as meaning “velocity too low to measure.”

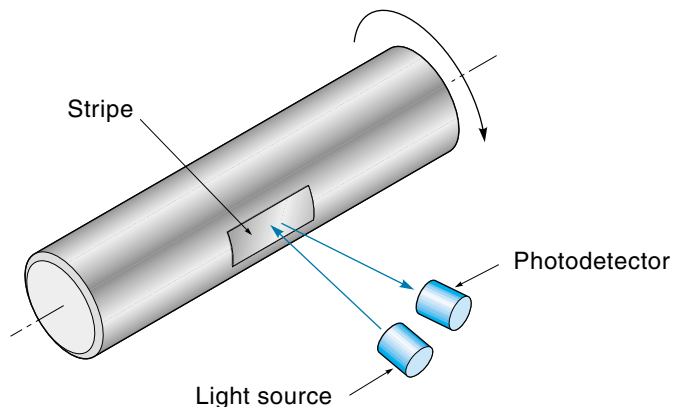
## Tachometers

### Optical Tachometers

The **optical tachometer**, a simple device, can determine a shaft speed in terms of revolutions per minute (rpm). As shown in Figure 6.23, a contrasting stripe is placed on the shaft. A photo sensor is mounted in such a way as to output a pulse each time the stripe goes by. The period of this waveform is inversely proportional to the rpm of the shaft and can be measured using a counter circuit like that described for the optical shaft

**Figure 6.23**

An optical tachometer.



encoder (Figure 6.22). Notice that this system cannot sense position or direction. However, if two photo sensors are used, the direction could be determined by phasing, similar to the incremental optical shaft encoder.

### Toothed-Rotor Tachometers

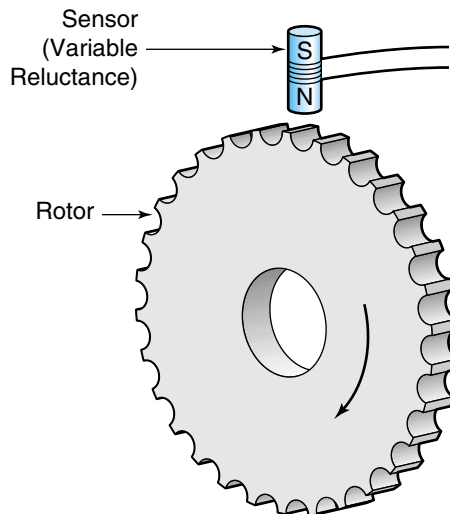
A **toothed-rotor tachometer** consists of a stationary sensor and a rotating, toothed, iron-based wheel (see Figure 6.24). The toothed wheel (which looks like a big gear) can be built into the part to be measured—for example, the crankshaft of a car engine. The sensor generates a pulse each time a tooth passes by. The angular velocity of the wheel is proportional to the frequency of the pulses. For example, if the wheel had 20 teeth, then there would be 20 pulses per revolution (see Example 6.10).

There are two kinds of toothed-rotor sensors in use. One kind is called a **variable-reluctance sensor** and consists of a magnet with a coil of wire around it (see Figure 6.24). Each time an iron tooth passes near the magnet, the magnetic field within the magnet increases, inducing a small voltage in the coil of wire. These pulses can be converted into a clean square wave with threshold detector circuits (as discussed in Chapter 3). The other type of sensor used for this application is the Hall-effect sensor. The details of the Hall-effect sensor are discussed later in this chapter, so for now we will simply say that this sensor also gives a pulse each time an iron tooth passes by.

#### EXAMPLE 6.10

A toothed-rotor sensor has 20 teeth. Find the revolutions per minutes (rpm) if the sensor outputs pulses at 120 Hz.

**Figure 6.24**  
A toothed-rotor  
tachometer.



### SOLUTION

One approach is to find the general transfer function (TF) for the system and then use the TF to find the rpm for any particular frequency. We start with the fact that 1 rps (1 revolution per second) of the rotor would result in a sensor frequency of 20 Hz.

$$\text{TF} = \frac{\text{output}}{\text{input}} = \frac{\text{freq (Hz)}_{\text{sensor}}}{\text{rpm}_{\text{rotor}}} = \frac{20 \text{ Hz}}{1 \text{ rps}} \times \frac{1 \text{ rps}}{60 \text{ rpm}} = \frac{0.33 \text{ Hz}_{\text{sensor}}}{1 \text{ rpm}_{\text{rotor}}}$$

Therefore, the input/output relationship of the system is: 1 rpm of the rotor produces a frequency of .33 Hz at the sensor. We can use this relationship to find the rpm of the rotor when the sensor frequency was 120 Hz.

$$120 \text{ Hz} \times \frac{1 \text{ rpm}}{0.33 \text{ Hz}} = 360 \text{ rpm}_{\text{rotor}}$$

Thus the rotor is turning at 360 rpm. Notice that we had to invert the TF in this case to get the units of Hz to cancel, leaving the desired unit of rpm.

### *Direct Current Tachometers*

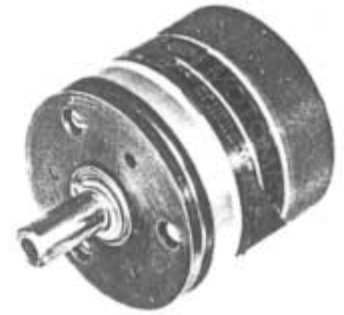
A **direct current tachometer** is essentially a DC generator that produces a DC output voltage proportional to shaft velocity. The output polarity is determined by the direction of rotation. Typically, these units have stationary permanent magnets (discussed in Chapter 7), and the rotating part consists of coils. Such a design keeps the inertia down but requires the use of brushes, which eventually wear out. Still, these units are useful because they provide a direct conversion between velocity and voltage.

Figure 6.25 gives the specifications of the CK20 tachometer. The housing of this unit is constructed so that it can mount “piggyback” on a motor, providing direct feedback of the motor velocity. The transfer function for the tachometer has units of volts/1000 rpm. We can use the transfer function to calculate the output voltage for a particular speed. Looking at the bottom of Figure 6.25, you can see that the CK20 comes in three models. For example, the CK20-A outputs 3 V for 1000 rpm (3 V/Krpm). It has a speed range of 0-6000 rpm, so the maximum voltage would be 18 V at 6000 rpm. This information can be displayed as a linear graph (Figure 6.26). From the graph, we can easily find the output voltage for any speed. The “linearity” of the motor is given as 0.2%, which means that the actual velocity may be as much as 0.2%, different from what it should be. For example, if the output is 9 V, the velocity should be 3000 rpm; however, because  $0.2\% \times 3000 = 6 \text{ rpm}$ , the actual velocity could be anywhere from 2994 to 3006 rpm.

**Figure 6.25**  
The CK20 DC  
tachometer.

## CK20 DC TACHOMETER

The model CK20 is a moving coil tachometer designed for use in applications requiring velocity feedback with minimum system inertia load.



<u>Parameter</u>	<u>Value</u>	<u>Units</u>
Linerity	.2	% max. deviation
Ripple	1.5	max, % peak to peak AC
Ripple Frequency	19	Cycles per revolution
Speed Range	1-6000	RPM
Armature Inertia	$9 \times 10^{-5}$	in-oz-sec <sup>2</sup>
Friction Torque	.25	in-oz, max.
Rated Life	10,000	Hours at 3000 RPM

### WINDING VARIATIONS

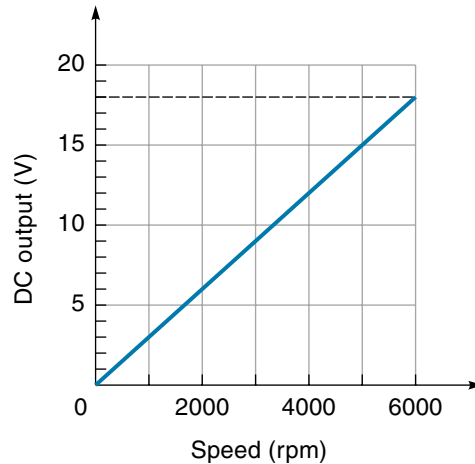
	<u>CK20-A</u>	<u>CK20-B</u>	<u>CK20-C</u>
Output Voltage Gradient (V/KRPM)	3.0	2.5	1

Velocities of thousands of rpm are much higher than you would normally find for actual heavy mechanical parts. Therefore the tachometer is frequently attached to the motor, and the motor is geared down to drive the load. Example 6.11 demonstrates this.



**Figure 6.26**

Graph of speed versus output DC volts for the CK20-A tachometer.

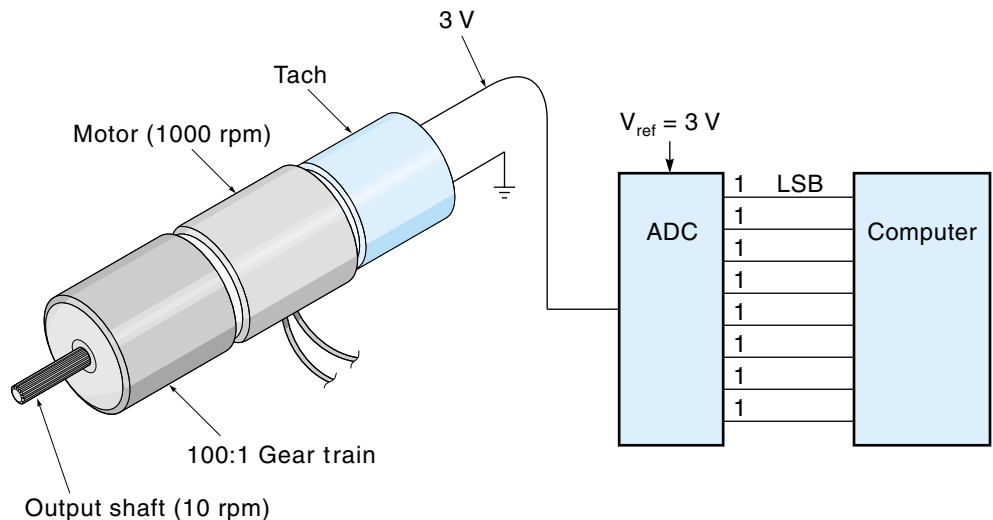
**EXAMPLE 6.11**

As shown in Figure 6.27, a motor with a piggyback tachometer has a built-in gear box with a ratio of 100 : 1 (that is, the output shaft rotates 100 times slower than the motor). The tachometer is a CK20-A with an output of 3 V/Krpm. This unit is driving a machine tool with a maximum rotational velocity of  $60^\circ/s$ .

- What is the expected output of the tachometer?
- Find the resolution of this system if the tachometer data were converted to digital with an 8-bit ADC as illustrated in Figure 6.27.

**Figure 6.27**

A tachometer interface circuit (Example 6.11).



**SOLUTION**

- a. A maximum tool velocity of  $60^\circ/\text{s}$  can be converted to rpm as follows:

$$\frac{60^\circ}{\text{s}} \times \frac{1 \text{ rev}}{360^\circ} \times \frac{60 \text{ s}}{\text{min}} = 10 \text{ rpm}$$

Because of the gear ratio, the tachometer is turning 100 times faster than the tool. Calculating the overall transfer function of the velocity sensor, we find

$$\underbrace{\frac{3 \text{ V}}{1000 \text{ rpm}_{\text{motor}}}}_{\text{Tachometer}} \times \underbrace{\frac{100 \text{ rpm}_{\text{motor}}}{1 \text{ rpm}_{\text{tool}}}}_{\text{Gearbox}} = 0.3 \text{ V/rpm}_{\text{tool}}$$

Now, using this transfer function, we can calculate what the tachometer voltage would be when the tool is rotating at 10 rpm:

$$V_{\text{tach}} = \frac{0.3 \text{ V}}{\text{rpm}_{\text{tool}}} \times 10 \text{ rpm}_{\text{tool}} = 3 \text{ V}$$

- b. To get the best resolution, we would reference the ADC to 3 V so that  $3 \text{ V} = 11111111_{\text{bin}}$  (255 decimal). Because we know that the tachometer is producing 3 V when the shaft is rotating at 10 rpm and that 8 bits represent 255 levels, we can calculate the rpm represented by each binary state:

$$\text{Resolution (LSB)} = \frac{10 \text{ rpm}}{255 \text{ states}} = 0.04 \text{ rpm/state}$$

This means that the digital controller will know the shaft velocity to within 0.04 rpm. Therefore, the resolution is 0.04 rpm.

**6.3 PROXIMITY SENSORS****Limit Switches**

A **proximity sensor** simply tells the controller whether a moving part is at a certain place. A **limit switch** is an example of a proximity sensor. A limit switch is a mechanical push-button switch that is mounted in such a way that it is actuated when a mechanical part or lever arm gets to the end of its intended travel. For example, in an automatic garage-door opener, all the controller needs to know is if the door is all the way open

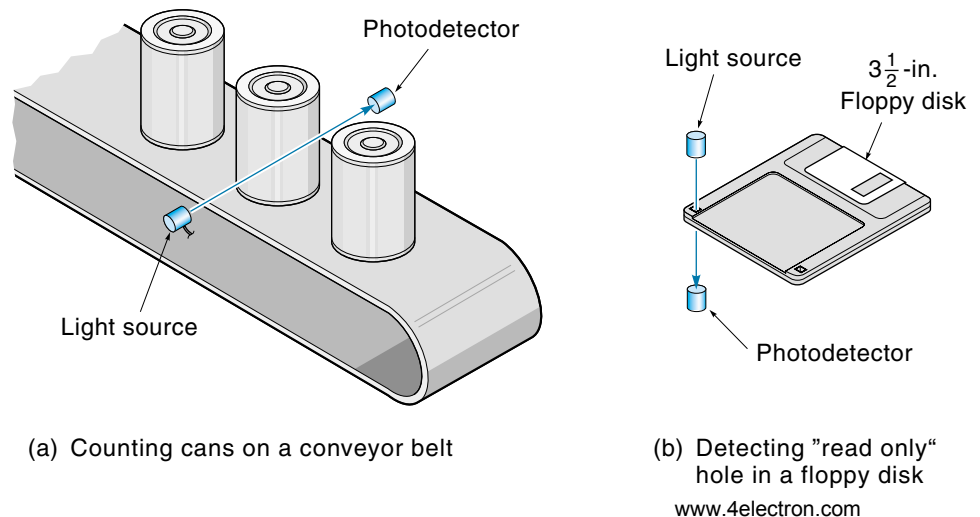
or all the way closed. Limit switches can detect these two conditions. Switches are fine for many applications, but they have at least two drawbacks: (1) Being a mechanical device, they eventually wear out, and (2) they require a certain amount of physical force to actuate. (Chapter 4 has more on limit switches.) Two other types of proximity sensors, which use either optics or magnetics to determine if an object is near, do not have these problems. The price we pay for these improved characteristics is that they require some support electronics.

## Optical Proximity Sensors

*Optical proximity sensors*, sometimes called *interrupters*, use a light source and a photo sensor that are mounted in such a way that the object to be detected cuts the light path. Figure 6.28 illustrates two applications of using photodetectors. In Figure 6.28(a), a photodetector counts the number of cans on an assembly line; in Figure 6.28(b), a photodetector determines whether the read-only hole in a floppy disk is open or closed. Optical proximity sensors frequently use a reflector on one side, which allows the detector and light source to be housed in the same enclosure. Also, the light source may be modulated to give the beam a unique “signature” so that the detector can distinguish between the beam and stray light.

Four types of photodetectors are in general use: photo resistors, photodiodes, photo transistors, and photovoltaic cells. A **photo resistor**, which is made out of a material such as cadmium sulfide (CdS), has the property that its resistance decreases when the light level increases. It is inexpensive and quite sensitive—that is, the resistance can change by a factor of 100 or more when exposed to light and dark.

**Figure 6.28**  
Two applications  
of a photo detector.



**Figure 6.29**  
Photodectors.

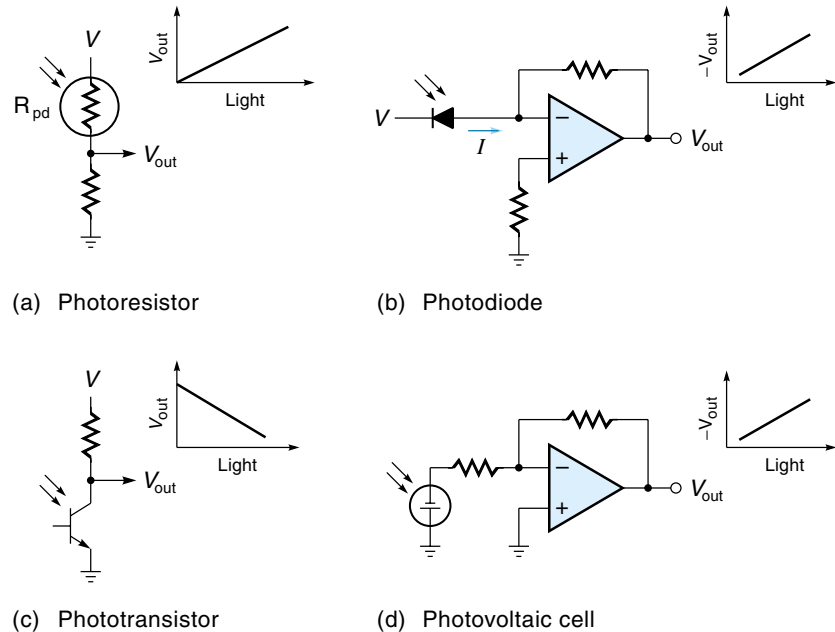


Figure 6.29(a) shows a typical interface circuit—as the light increases,  $R_{pd}$  decreases, and so  $V_{out}$  increases.

A **photodiode** is a light-sensitive diode. A little window allows light to fall directly on the PN junction where it has the effect of increasing the reverse-leakage current. Figure 6.29(b) shows the photodiode with its interface circuit. Notice that the photodiode is reversed-biased and that the small reverse-leakage current is converted into an amplified voltage by the op-amp.

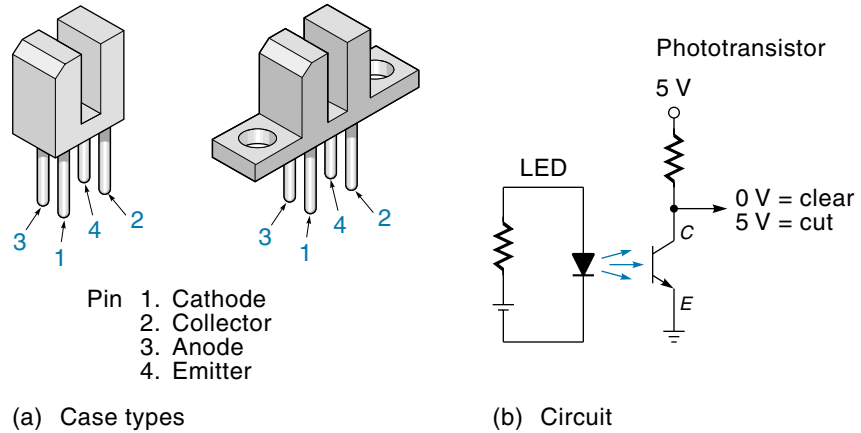
A **photo transistor** [Figure 6.29(c)] has no base lead. Instead, the light effectively creates a base current by generating electron-hole pairs in the CB junction—the more light, the more the transistor turns on.

The **photovoltaic cell** is different from the photo sensors discussed so far because it actually creates electrical power from light—the more light, the higher the voltage. (A solar cell is a photovoltaic cell.) When used as a sensor, the small voltage output must usually be amplified, as shown in Figure 6.29(d).

Some applications make use of an optical proximity sensor called a **slotted coupler**, also called an *optointerrupter* (Figure 6.30). This device includes the light source and detector in a single package. When an object moves into the slot, the light path is broken. The unit comes in a wide variety of standard housings [Figure 6.30(a)]. To operate, power must be provided to the LED, and the output signal taken from the phototransistor. This is done in the circuit of Figure 6.30(b), which provides a TTL-level (5 V or 0 V) output. When the slot is open, the light beam strikes the transistor, turning it on, which grounds the collector. When the beam is interrupted, the transistor turns off, and the collector is pulled up to 5 V by the resistor.

**Figure 6.30**

An optical slotted coupler.



Optical sensors enjoy the advantage that neither the light source, the object to be detected, nor the detector have to be near each other. An example of this is a burglar alarm system. The light source is on one side of the room, the burglar is in the middle, and the detector is on the other side of the room. This property can be important in a case where there are no convenient mounting surfaces near the part to be measured. On the other hand, keeping the lenses clean may be a problem in some industrial situations.

### Hall-Effect Proximity Sensors

In 1879 E. H. Hall first noticed the effect that bears his name. He discovered a special property of copper, and later of semiconductors: They produce a voltage in the presence of a magnetic field. This is especially true for germanium and indium. The **Hall effect**, as it is called, was originally used for wattmeters and gaussmeters; now it is used extensively for proximity sensors. Figure 6.31 shows some typical applications. In all cases, the Hall-effect sensor outputs a voltage when the magnetic field in which it finds itself increases. This is done either by moving a magnet or by changing the magnetic field path (but the value of the Hall voltage does not depend on the field “moving”—only on the field being there).

Figure 6.32 shows how the Hall effect works. First, an external voltage source is used to establish a current ( $I$ ) in the semiconductor crystal. The output voltage ( $V_H$ ) is sensed across the sides of the crystal, perpendicular to the current direction. When a magnetic field is brought near, the negative charges are deflected to one side producing a voltage. The relationship can be described in the following equation:

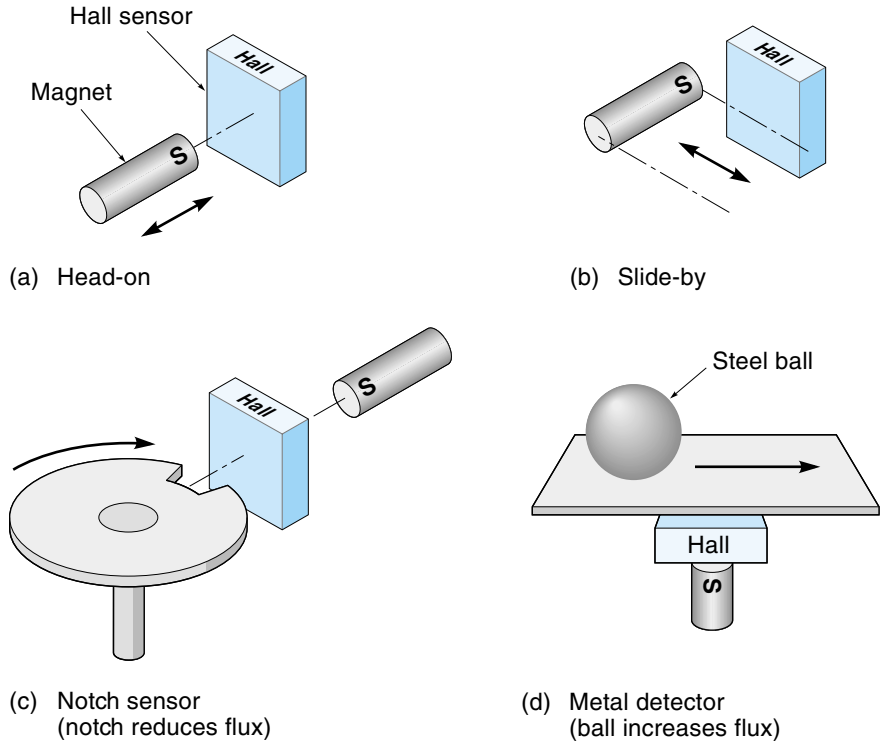
$$V_H = \frac{KIB}{D} \quad (6.4)$$

where

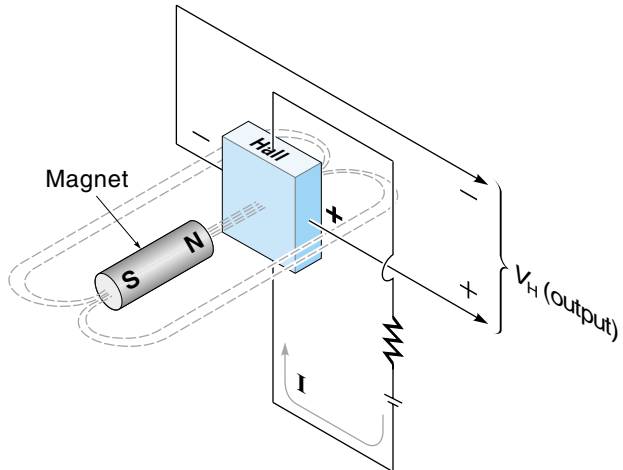
$V_H$  = Hall-effect voltage

$K$  = constant (dependent on material)

**Figure 6.31**  
Typical applications  
of Hall-effect sensors.

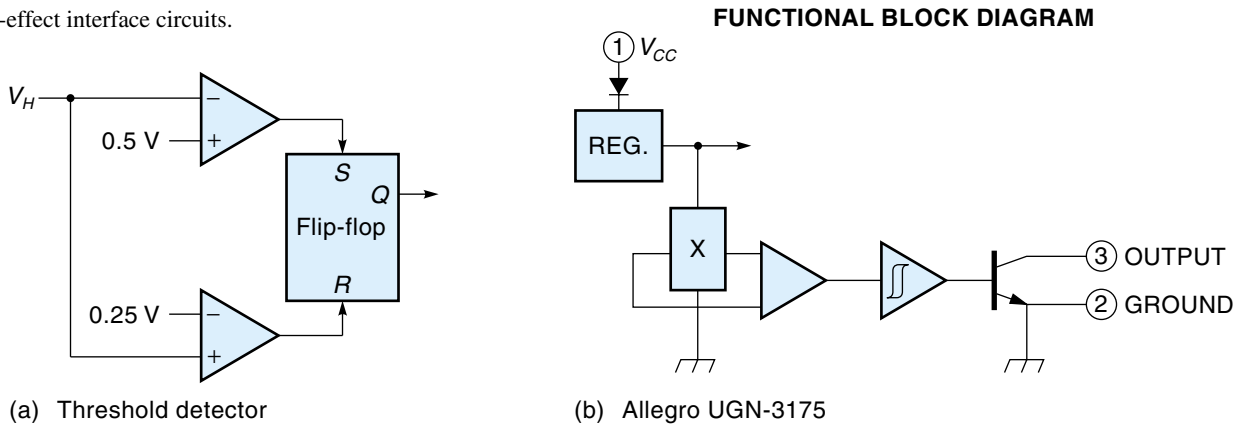


**Figure 6.32**  
The operation of a  
Hall-effect sensor.



**Figure 6.33**

Hall-effect interface circuits.



$I$  = current from an external source

$B$  = magnetic flux density

$D$  = thickness constant

Equation 6.4 states that  $V_H$  is directly proportional to  $I$  and  $B$ . If  $I$  is held constant, then  $V_H$  is directly proportional to  $B$  (magnetic flux density). Therefore, the output is not really on/off but (over a short distance) somewhat linear. To get a switching action, the output must go through a **threshold detector** like that illustrated in Figure 6.33(a). This circuit uses two comparator amps to establish the high and low switching voltages. When  $V_H$  goes above 0.5 V, the top amp sets the  $R$ - $S$  flip-flop. When  $V_H$  goes below 0.25 V, the bottom amp resets the flip-flop. For this circuit to work, we need to make sure that the magnet comes near enough to the sensor to make  $V_H$  go above 0.5 V and far enough away for  $V_H$  to drop below 0.25 V.

A complete Hall-effect switch can be purchased in IC form. One example is the Allegro 3175 [Figure 6.32(b)]; it includes the sensor ( $X$ ), the cross-current drive, and the threshold detector. The transistor turns on when the magnetic field goes above +100 gauss and turns off when the field drops below -100 gauss. The transistor can sink 15 mA, which can drive a small relay directly or a TTL digital circuit.

Hall-effect sensors are used in many applications—for example, computer keyboard switches and proximity sensors in machines. They are also used as the sensors in the toothed-rotor tachometers discussed earlier in this chapter.

## 6.4 LOAD SENSORS

*Load sensors* measure mechanical force. The forces can be large or small—for example, weighing heavy objects or detecting low-force tactile pressures. In most cases, it is the slight deformation caused by the force that the sensor measures, not the force directly.

Typically, this deformation is quite small. Once the amount of tension (stretching) or compression (squeezing) displacement has been measured, the force that must have caused it can be calculated using the mechanical parameters of the system. The ratio of the force to deformation is a constant for each material, as defined by **Hooke's law**:

$$F = KX \quad (6.5)$$

where

$K$  = spring constant of the material

$F$  = applied force

$X$  = extension or compression as result of force

For example, if a mechanical part has a spring constant of 1000 lb/in. and it compresses 0.5 in. under the load, then the load must be 500 lb.

### Bonded-Wire Strain Gauges

The **bonded-wire strain gauge** can be used to measure a wide range of forces, from 10 lb to many tons. It consists of a thin wire (0.001 in.) looped back-and-forth a few times and cemented to a thin paper backing [Figure 6.34(a)]. More recent versions use printed-circuit technology to create the wire pattern. The entire strain gauge is securely bonded to some structural object and will detect any deformation that may take place. The gauge is oriented so the wires lie in the same direction as the expected deformation. The principle of operation is as follows: If the object is put under tension, the gauge will stretch and elongate the wires. The wires not only get slightly longer but also thinner. Both actions cause the total wire resistance to rise, as can be seen from the basic resistance equation:

$$R = \frac{\rho L}{A} \quad (6.6)$$

where

$R$  = resistance of a length of wire (at 20°C)

$\rho$  = resistivity (a constant dependent on the material)

$L$  = length of wire

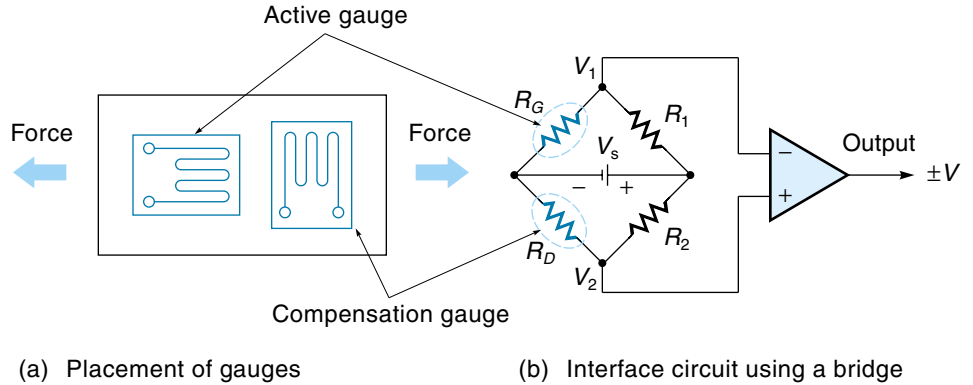
$A$  = cross-sectional area of wire

The change in resistance of the strain-gauge wires can be used to calculate the elongation of the strain gauge (and the object to which it is cemented). If you know the elongation and the spring constant of the supporting member, then the principles of Hooke's law can be used to calculate the force being applied.

The resistance change in a strain gauge is small. Typically, it is only a few percent, which may be less than an ohm. Measuring such small resistances usually requires a bridge circuit [Figure 6.34(b)]. With this circuit, a small change in one resistor can cause a relatively large percentage change in the voltage across the bridge. Initially, the bridge is balanced (or "nulled") by adjusting the resistances so that  $V_1 = V_2$ . Then, when the



**Figure 6.34**  
Strain gauges.



gauge resistance changes, the voltage difference ( $V_1 - V_2$ ) changes. The bridge also allows us to cancel out variations due to temperature, by connecting a **compensating gauge** (known as the *dummy*) as one of the bridge resistors. As shown in Figures 6.34 and 6.35, the actual compensation gauge is placed physically near the active gauge so as to receive the same temperature, but it is oriented perpendicularly from the active gauge so the force will not elongate its wires.

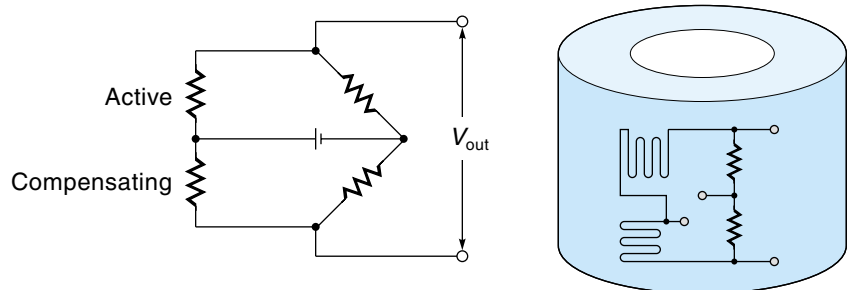
Analyzing the bridge circuit of Figure 6.34, we first calculate the individual voltages  $V_1$  and  $V_2$  using the voltage divider law:

$$V_1 = \frac{V_s R_G}{R_1 + R_G} \qquad V_2 = \frac{V_s R_D}{R_2 + R_D}$$

The voltage across the bridge can be expressed as ( $V_1 - V_2$ )

$$(V_1 - V_2) = \Delta V = V_s \left( \frac{R_G}{R_1 + R_G} - \frac{R_D}{R_2 + R_D} \right)$$

**Figure 6.35**  
Strain-gauge configurations.



(a) Active and compensating gauges are placed together so that they will be at the same temperature

(b) Load cell with strain gauge and bridge

Using algebra, we can convert this equation to

$$\Delta V = V_s \frac{(R_G R_2 - R_D R_1)}{(R_1 + R_G)(R_2 + R_D)}$$

We can simplify the analysis by specifying that all the resistors in the bridge (including  $R_G$  and  $R_D$ ) have the same value ( $R$ ) when it is balanced. Then, when the gauge is stretched,  $R_G$  will increase a little to become  $R + \Delta R$  (where  $\Delta R$  is how many ohms  $R_G$  increased because of the stretching). Using these conditions, the equation above simplifies to

$$\Delta V = V_s \frac{\Delta R}{4R + 2\Delta R} \quad (\text{when all resistors in bridge} = R \text{ at null})$$

Looking at the denominator, we see that it is the sum of  $4R$  and  $2\Delta R$ , but in all realistic situations  $4R$  will be much, much larger than  $2\Delta R$ , so we could say that  $4R + 2\Delta R \approx 4R$ . With this assumption and some algebraic rearranging, we arrive at Equation 6.7, which we can use to calculate the change in strain-gauge resistance on the basis of measured voltage change across the bridge.

$$\Delta R \approx \frac{4R \Delta V}{V_s} \quad (6.7)$$

where

$\Delta R$  = change in the strain-gauge resistance

$R$  = nominal value of all bridge resistors

$\Delta V$  = voltage detected across the bridge

$V_s$  = source voltage applied to the bridge

As the strain gauge is stretched, its resistance rises. The precise relationship between elongation and resistance can be computed using Equation 6.8 and is based on the **gauge factor** (GF), which is supplied by the strain-gauge manufacturer:

$$\varepsilon = \frac{\Delta R/R}{\text{GF}} \quad (6.8)$$

where

$\varepsilon$  = elongation of the object per unit of length ( $\Delta L/L$ ), called *strain*

$R$  = strain-gauge resistance

$\Delta R$  = change in strain-gauge resistance due to force

GF = gauge factor, a constant supplied by the manufacturer (GF is the ratio  $(\Delta R/R)/(\Delta L/L)$ )

One more equation is needed before we can solve a strain-gauge problem—an equation that relates *stress* and the resulting *strain* in an object. **Stress** is the force per cross-sectional area; for example, if a table leg has a cross-sectional area of 2 in<sup>2</sup> and is supporting a load of 100 lb, then the stress is 50 lb/in<sup>2</sup>. **Strain** is the amount of length (per unit length) that the object stretches as a result of being subjected to a stress; for

example, if an object 10 in. long stretches 1 in., then each inch of the object stretched 0.1 in., and so the strain would be 0.1 in./in. Stress and strain are related by a constant called **Young's modulus** (also called *modulus of elasticity*), as shown in Equation 6.9. Young's modulus ( $E$ ) is a measure of how stiff a material is and could be thought of as a kind of spring constant:

$$E = \frac{\rho}{\epsilon} \quad (6.9)$$

where

$E$  = Young's modulus (a constant for each material)

$\rho$  = stress (force per cross-sectional area)

$\epsilon$  = strain (elongation per unit length)

Table 6.2 gives some values of  $E$  for common materials.

### EXAMPLE 6.12

A strain gauge and bridge circuit are used to measure the tension force in a steel bar (Figure 6.36). The steel bar has a cross-sectional area of 2 in<sup>2</sup>. The strain gauge has a nominal resistance of 120  $\Omega$  and a GF of 2. The bridge is supplied with 10 V. When the bar is unloaded, the bridge is balanced so the output is 0 V. Then force is applied to the bar, and the bridge voltage goes to 0.0005 V. Find the force on the bar.

### SOLUTION

First, use Equation 6.7 to calculate the change in strain-gauge resistance due to the applied force:

$$\Delta R \approx \frac{4R\Delta V}{V_S} = \frac{4 \times 120 \Omega \times 0.0005 \text{ V}}{10 \text{ V}} = 0.024 \Omega$$

**TABLE 6.2**

### Young's Modulus ( $E$ ) for Common Materials

Substance	lb/in	N/cm <sup>2</sup>
Steel	$30 \times 10^6$	$2.07 \times 10^7$
Copper	$15 \times 10^6$	$1.07 \times 10^7$
Aluminum	$10 \times 10^6$	$6.9 \times 10^6$
Rock	$7.3 \times 10^6$	$5.0 \times 10^6$
Hard wood	$1.5 \times 10^6$	$1.0 \times 10^6$

Next, use Equation 6.8 to calculate the elongation (strain) of the strain gauge (how much it was stretched):

$$\varepsilon = \frac{\Delta R/R}{GF} = \frac{0.024/120}{2} = 0.0001 \text{ in./in.}$$

Finally, use Equation 6.9

$$E = \frac{\rho}{\varepsilon}$$

to calculate the force on the bar. This will require looking up the value of Young's modulus. From Table 6.2, we find it to be 30,000,000 lb/in<sup>2</sup> for steel. Rearranging Equation 6.9 gives

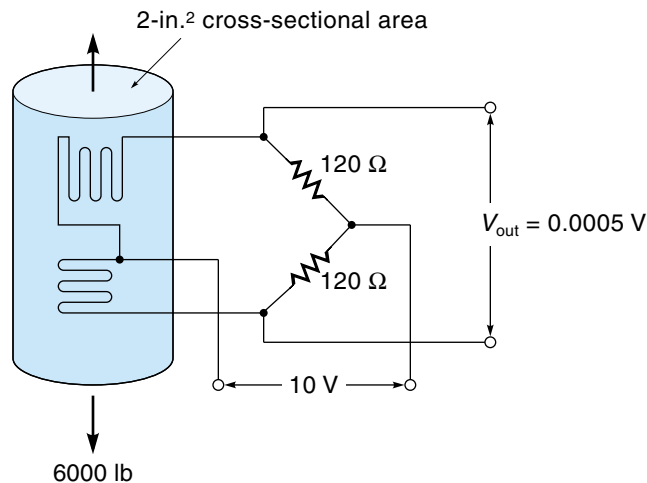
$$\rho = E\varepsilon = 30,000,000 \text{ lb/in}^2 \times 0.0001 \text{ in./in.} = 3000 \text{ lb/in}^2$$

This result tells us that the tension force on the steel bar is 3000 lb/in<sup>2</sup>, and because this bar has a cross-sectional area of 2 in<sup>2</sup>, the total tension force in the bar is 6000 lb.

### EXAMPLE 6.12 (Repeated with SI Units)

A strain gauge and bridge circuit are used to measure the tension force in a bar of steel that has a cross-sectional area of 13 cm<sup>2</sup>. The strain gauge has a nominal resistance of 120 Ω and a GF of 2. The bridge is supplied with 10 V.

**Figure 6.36**  
Strain-gauge measuring tension in steel bar (Example 6.12).



When the bar is unloaded, the bridge is balanced so the output is 0 V. Then force is applied to the bar, and the bridge voltage goes to 0.0005 V. Find the force on the bar.

### SOLUTION

First, calculate the change in strain-gauge resistance due to the applied force:

$$\Delta R \approx \frac{4R\Delta V}{V_S} = \frac{4 \times 120 \, \Omega \times 0.0005 \, \text{V}}{10 \, \text{V}} = 0.024 \, \Omega$$

Next, calculate the elongation (strain) of the strain gauge:

$$\varepsilon = \frac{\Delta R/R}{GF} = \frac{0.024/120}{2} = 0.0001 \, \text{cm/cm}$$

Finally, use Equation 6.9

$$E = \frac{\rho}{\varepsilon}$$

to calculate the force on the bar. This will require looking up the value of Young's modulus. From Table 6.2, we find it to be  $2.07 \times 10^7 \, \text{N/cm}^2$  for steel. Rearranging Equation 6.9 gives

$$\rho = E\varepsilon = 20,700,000 \, \text{N/cm}^2 \times 0.0001 \, \text{cm/cm} = 2070 \, \text{N/cm}^2$$

This result tells us that the tension force on the steel bar is  $2070 \, \text{N/cm}^2$ , and because this bar has a cross-sectional area of  $13 \, \text{cm}^2$ , the total tension force in the bar is 26,910 N.

Strain-gauge force transducers (called *load cells*) are available as self-contained units that can be mounted anywhere in the system. A load cell may contain two strain gauges (active and compensating) and a bridge [Figure 6.35]. A typical application for load cells is monitoring the weight of a tank. The tank would be sitting on three or four load cells, so the weight of the tank is the sum of the outputs of the load cells [see Figure 6.60(c)].

### Semiconductor Force Sensors

Another type of force sensor uses the **piezoresistive effect** of silicon. These units change resistance when force is applied and are 25-100 times more sensitive than the bonded-wire strain gauge. A semiconductor strain gauge is a single strip of silicon material that is bonded to the structure. When the structure stretches, the silicon is elongated, and the resistance from end to end increases (however, the resistance change is nonlinear).

## Low-Force Sensors

Some applications call for low-force sensors. For example, imagine the sensitivity required for a robot gripper to hold a water glass without slipping and without crushing it. Strain gauges can measure low forces if they are mounted on an elastic substrate, like rubber—then a small force will cause a significant deflection and resistance change. Another solution would be to construct a low-force sensor with a spring and a linear-motion potentiometer (Figure 6.37). The spring compresses a distance proportional to the applied force, and this distance is measured with the pot.

### EXAMPLE 6.13

Construct a force sensor with the following characteristics;

Range: 0-30 lb

Deformation: 0.5 in. (maximum)

Output: 0.1 V/lb

A 1 k $\Omega$  linear motion pot is available with a 1-in. stroke.

### SOLUTION

Using the concept of Figure 6.37, we first need to specify the spring. The specifications call for a spring that deforms 0.5 in. with 30 lb of force. Thus,

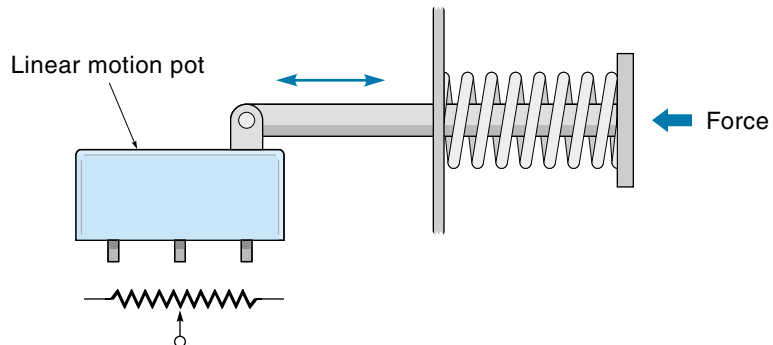
$$K \text{ (spring constant)} = \frac{30 \text{ lb}}{0.5 \text{ in.}} = 60 \text{ lb/in.}$$

Knowing we need a spring with a  $K$  of 60 lb/in., we could go to a spring catalog and select one.

The desired sensitivity of 0.1 V/lb dictates that the maximum output voltage will be 3 V when the force is 30 lb: voltage at maximum load = 30 lb  $\times$  0.1 V/lb = 3 V.

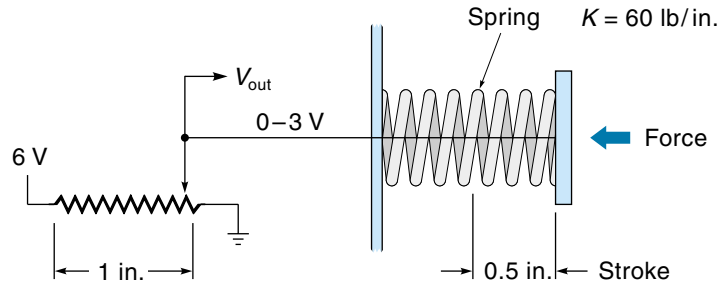
**Figure 6.37**

A tactile force sensor using a spring-loaded linear pot.



**Figure 6.38**

A tactile sensor setup  
(Example 6.13).



Finally, we must determine the supply voltage across the pot. The pot should output 3 V when it is moved 0.5 in. (one-half of its stroke). A ratio can be used to find the pot supply voltage for 1 in. of stroke:

$$\frac{3 \text{ V}}{0.5 \text{ in.}} = \frac{X}{1 \text{ in.}}$$

$$X = 6 \text{ V}$$

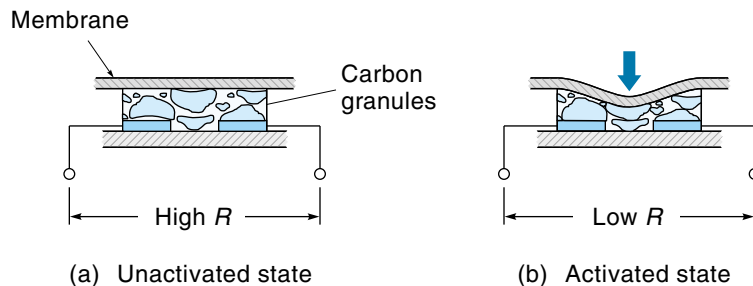
Therefore, the supply voltage should be 6 V. Figure 6.38 shows the final setup.

A very low-force tactile sensor can be made using conductive foam. This is the principle used in membrane keypads illustrated in Figure 6.39. The conductive foam is a soft foam rubber saturated with very small carbon particles. When the foam is squeezed, the carbon particles are pushed together, and the resistance of the material falls. Therefore, in some fashion, resistance is proportional to force. At present this concept has found limited application in such things as calculator keypads; because of its simplicity and low cost, however, it is a viable option for other applications such as robot tactile sensors.

Finally, a very simple tactile sensor can be made with two or more limit switches mounted side-by-side with spring actuators that are set to switch at different pressures.

**Figure 6.39**

Conductive-foam  
tactile sensor.



As the pressure increases, the first switch closes, then with more pressure the next switch closes, and so on.

## 6.5 PRESSURE SENSORS

*Pressure* is defined as the force per unit area that one material exerts on another. For example, consider a 10-lb cube resting on a table. If the area of each face of the cube is 4 in<sup>2</sup>, then 10 lb is distributed over an area of 4 in<sup>2</sup>, so the cube exerts a pressure on the table of 2.5 lb/in<sup>2</sup> (10 lb/4 in<sup>2</sup> = 2.5 lb/in<sup>2</sup>, or 2.5 **psi**). In SI units, pressure is measured in Newtons per square meter (N/m<sup>2</sup>), which is called a Pascal (Pa). For a liquid, pressure is exerted on the side walls of the container as well as the bottom.

Pressure sensors usually consist of two parts: The first converts pressure to a force or displacement, and the second converts the force or displacement to an electrical signal. Pressure measurements are made only for gases and liquids. The simplest pressure measurement yields a **gauge pressure**, which is the difference between the measured pressure and ambient pressure. At sea level, ambient pressure is equal to atmospheric pressure and is assumed to be 14.7 psi, or 101.3 kiloPascals (kPa). A slightly more complicated sensor can measure **differential pressure**, the difference in pressure between two places where neither pressure is necessarily atmospheric. A third type of pressure sensor measures **absolute pressure**, which is measured with a differential pressure sensor where one side is referenced at 0 psi (close to a total vacuum).

### Bourdon Tubes

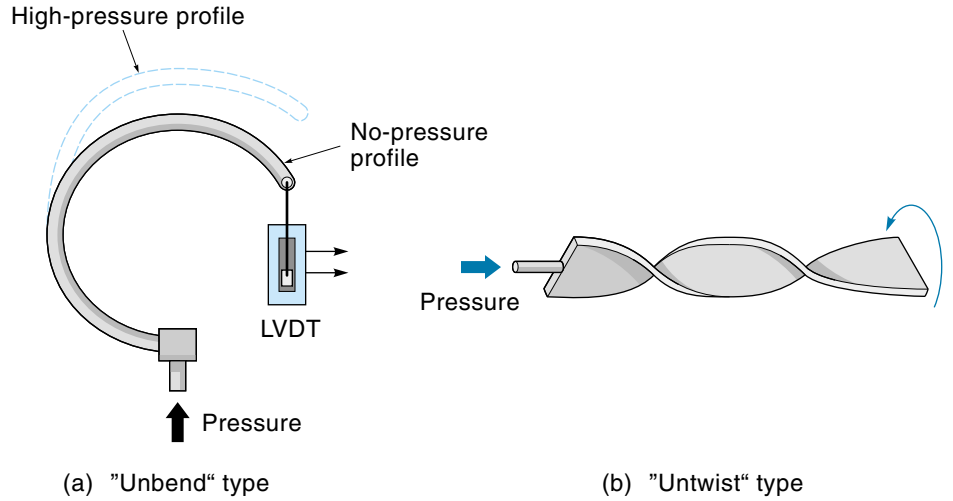
A **Bourdon tube** is a short bent tube, closed at one end. When the tube is pressurized, it tends to straighten out. This motion is proportional to the applied pressure. Figure 6.40 shows some Bourdon-tube configurations. Notice that the displacement can be either linear or angular. A position sensor such as a pot or LVDT can convert the displacement into an electrical signal. Bourdon-tube sensors are available in pressure ranges from 30 to 100,000 psi. Typical uses include steam- and water-pressure gauges.

### Bellows

This sensor uses a small metal bellows to convert pressure into linear motion [Figure 6.41(a)]. As the pressure inside increases, the bellows expand against the resistance of a spring (the spring is often the bellows itself). This motion is detected with a position sensor such as a pot. Figure 6.41(b) illustrates a differential pressure sensor, which can be made by enclosing the bellows in a canister. Here, the pressure from outside the bellows (pressure 2) tends to make the bellows compress, whereas pressure 1 tends to make the bellows expand. The position of the shaft is a function of the difference in pressure inside and outside the bellows. Bellows are capable of more sensitivity than the Bourdon tube in the lower-pressure range of 0-30 psi.



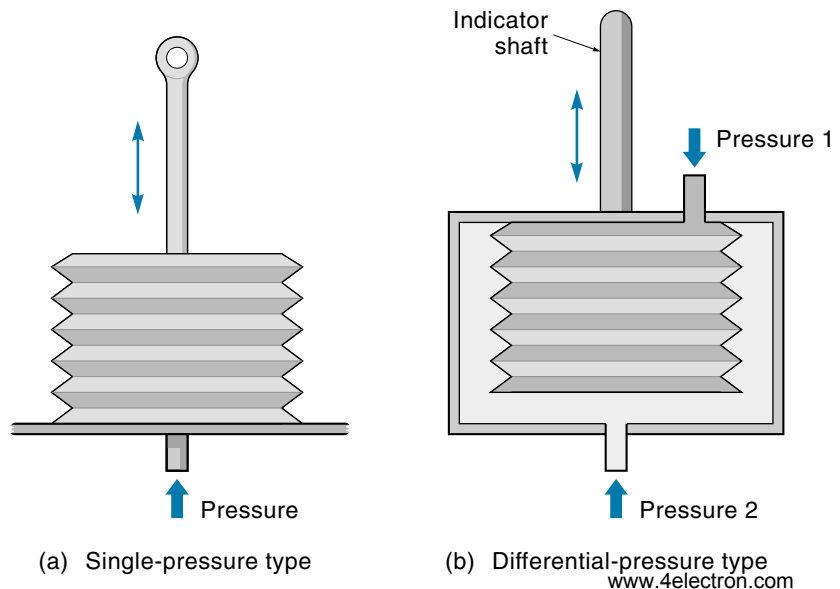
**Figure 6.40**  
Bourdon-tube sensors.



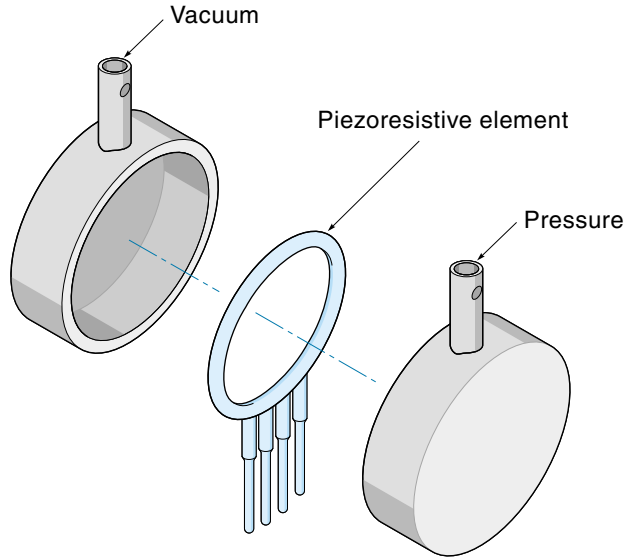
### Semiconductor Pressure Sensors

Some commercially available pressure sensors use the piezoresistive property of silicon (Figure 6.42). The piezoresistive element converts pressure directly into resistance, and resistance can be converted into voltage. These sensors have the advantage of “no moving parts” and are available in pressure ranges from 0-1.5 psi to 0-5000 psi. An example of a commercial semiconductor pressure sensor is the ST2000 series from Sen Sym Inc. (Figure 6.43). This unit can be used with fluids or gases, has an internal amplifier, and outputs a voltage that is directly proportional to absolute pressure.

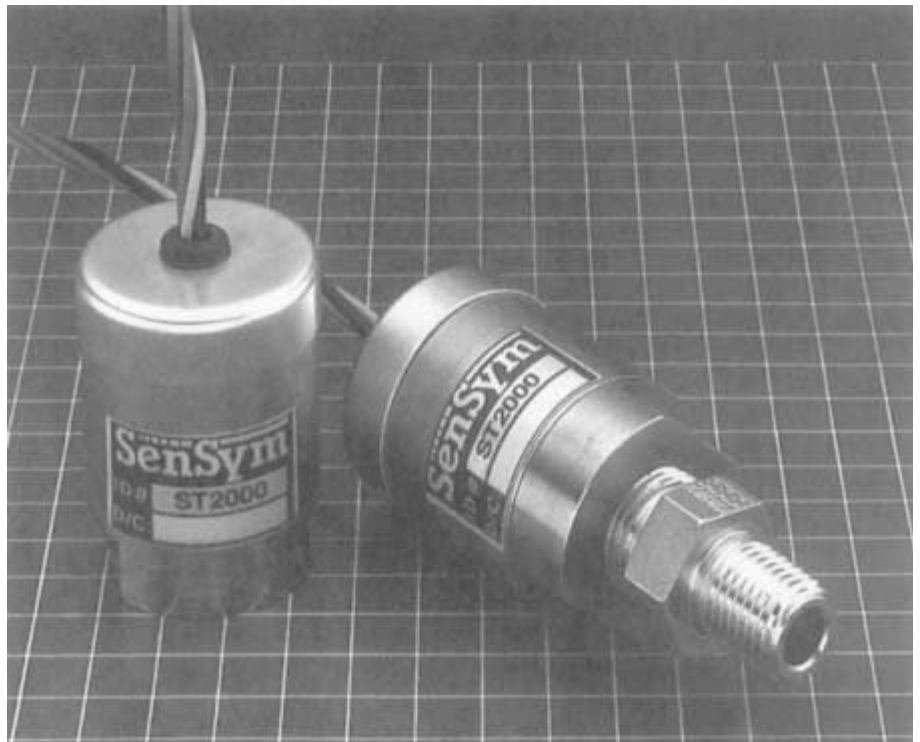
**Figure 6.41**  
Bellows pressure sensors.



**Figure 6.42**  
A semiconductor  
pressure sensor.



**Figure 6.43**  
The ST2000 semi-  
conductor pressure  
sensor. (Courtesy of  
SenSym Inc.)



## 6.6 TEMPERATURE SENSORS

*Temperature sensors* give an output proportional to temperature. Most temperature sensors have a *positive temperature coefficient* (desirable), which means that the sensor output goes up as the temperature goes up, but some sensors have a *negative temperature coefficient*, which means that the output goes down as the temperature goes up. Many control systems require temperature sensors, if only to know how much to compensate other sensors that are temperature-dependent. Some common types are discussed next.

### Bimetallic Temperature Sensors

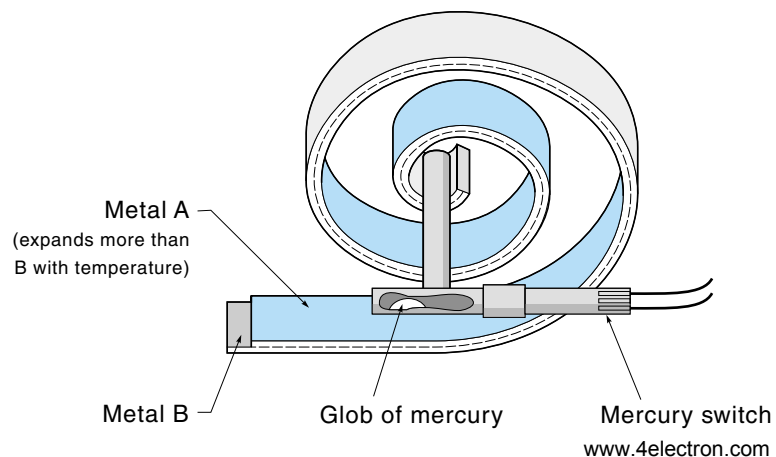
The **bimetallic temperature sensor** consists of a bimetallic strip wound into a spiral (Figure 6.44). The bimetallic strip is a laminate of two metals with different coefficients of thermal expansion. As the temperature rises, the metal on the inside expands more than the metal on the outside, and the spiral tends to straighten out. These sensors are typically used for on-off control as in a household thermostat where a mercury switch is rocked from on to off. In Figure 6.44, when the temperature increases, the tube containing liquid mercury rotates clockwise. When the tube rotates past the horizontal, the mercury runs down to the right and completes the electrical connection between the electrodes. One distinct advantage of this system is that the output from the switch can be used directly without further signal conditioning. Currently, mercury switches are being phased out because of environmental reasons, but contact-type switches are taking their place.

### Thermocouples

The **thermocouple** was developed over 100 years ago and still enjoys wide use, particularly in high-temperature situations. The thermocouple is based on the **Seebeck**

**Figure 6.44**

A bimetallic thermal sensor controlling a mercury switch (shown in “cold” state).



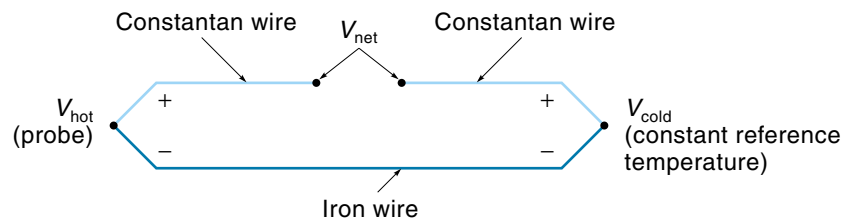
**effect**, a phenomenon whereby a voltage that is proportional to temperature can be produced from a circuit consisting of two dissimilar metal wires. For example, a thermocouple made from iron and constantan (an alloy) generates a voltage of approximately  $35 \mu\text{V}/^\circ\text{F}$ . Figure 6.45(a) illustrates this situation. We can think of the junctions at each end of the dissimilar metal wires as producing a voltage, so the net voltage ( $V_{\text{net}}$ ) is actually the difference between the junction voltages. One junction is on the probe and is called the **hot junction**. The other junction is kept at some known reference temperature and is called the **cold junction**, or *reference junction*. The output voltage from this system can be expressed as follows:

$$V_{\text{net}} = V_{\text{hot}} - V_{\text{cold}} \tag{6.10}$$

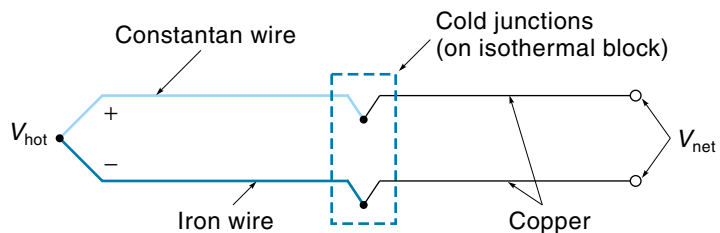
In practice, the thermocouple wires must connect to copper wires at some point as shown in Figure 6.45(b), so there are really three junctions. However, it turns out that the total voltage from the two copper junctions will be the same as the single cold-junction voltage ( $V_{\text{cold}}$ ) of Figure 6.45(a) (assuming the copper junctions are at the same temperature), so the analysis is unchanged.

Traditionally, the cold junction was kept at  $32^\circ\text{F}$  in an **ice-water bath**, which is water with ice in it. Ice water was used because it is one way to produce a known temperature, and so  $V_{\text{cold}}$  becomes a constant in Equation 6.10, leaving a direct relationship between  $V_{\text{net}}$  and  $V_{\text{hot}}$ :

**Figure 6.45**  
A Thermocouple circuit  
(iron–constantan type).



(a) Basic principle



(b) Thermocouple connected to copper wires

$$V_{\text{hot}} = V_{\text{net}} + \underbrace{\text{constant}}_{\substack{V_{\text{cold}} \text{ at } 32^{\circ}\text{F} \\ \text{or some other} \\ \text{reference temperature}}} \quad (6.10a)$$

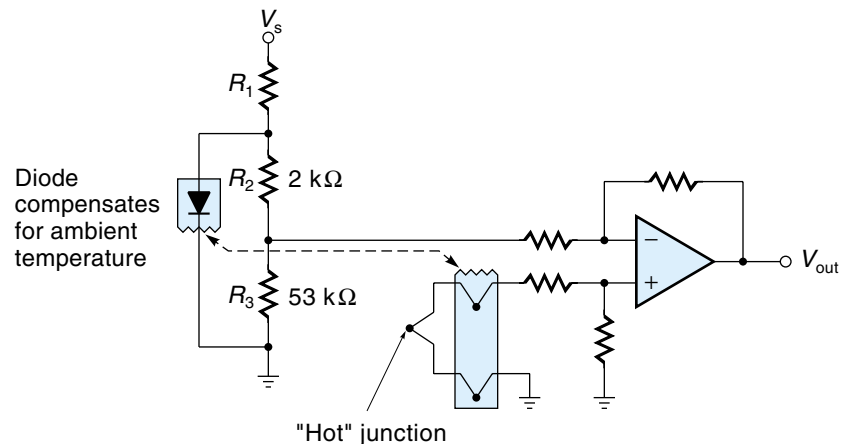
Modern systems eliminate the need for ice water. One method is to maintain the cold junction at constant temperature with a control system. This can be useful if there are many thermocouples in a system—they can all be referenced to the same temperature. Another method (used by a computer controller) is to simply look up in a table the value of  $V_{\text{cold}}$  for the ambient temperature and add this value to  $V_{\text{net}}$  to yield  $V_{\text{hot}}$ .

Still another way to eliminate the ice-water bath is to use a temperature-sensitive diode (in an interface circuit) that makes the thermocouple output behave as if the cold junction were still at freezing, even though it's not. Figure 6.46 shows such a circuit for an iron-constantan thermocouple. The cold junctions are maintained at the same temperature as the diode by mounting them all on an *isothermal* block. As the ambient temperature increases, the diode forward-voltage drop (about 0.6 V) decreases at a rate of about 1.1 mV/°F. This voltage is scaled down (with  $R_2$  and  $R_3$ ) to 28  $\mu\text{V}/^{\circ}\text{F}$ , which is the same rate that the real cold-junction voltage increases with ambient temperature. By subtracting (with the op-amp) the effect of the ambient temperature changes on the cold junction, we get a single thermocouple voltage that is directly proportional to temperature.

Commercial thermocouples are available with different temperature ranges and sensitivities (sensitivity being a measure of volts/degree). Figure 6.47 shows the volts versus temperature curves of the major classes of thermocouples. As you can see, type J (iron-constantan) has the highest sensitivity but the lowest temperature range, type K (chromel-alumel) has a higher temperature range but a lower sensitivity, and type R (platinum-rhodium) has an even lower sensitivity but can work at higher temperatures. Tables are available that give the precise thermocouple voltage-temperature relationship (for both °F and °C). See Appendix A for tables of type-J thermocouples.

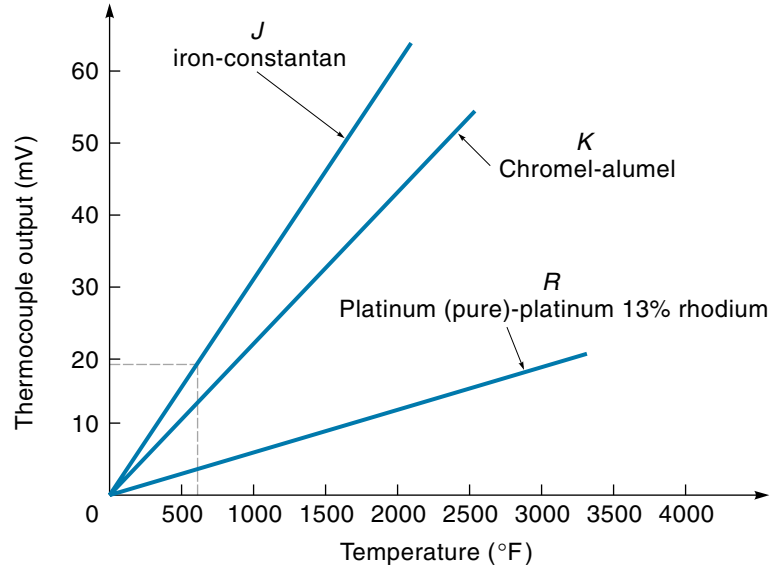
**Figure 6.46**

A diode being used to compensate for cold-junction voltage.



**Figure 6.47**

Thermocouple outputs for different wire types (referenced at 32°F).



Thermocouples are simple and rugged but require extra electronics to deal with the inherent low-sensitivity and cold-junction problems. However, because they are linear (over a limited range), reliable, and stable, they enjoy wide use in measuring high temperatures in furnaces and ovens.

#### EXAMPLE 6.14

An oven is supposed to be maintained at 1000°F by a control system, but you suspect that the temperature is much cooler. You have at your disposal a type-J thermocouple and a voltmeter. How would you use this equipment to check the oven temperature?

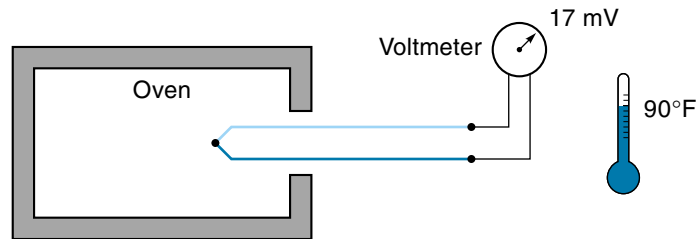
#### SOLUTION

First, put the thermocouple in the oven and connect the thermocouple leads to the voltmeter (Figure 6.48). Try to make the cold-junction connections to the meter probes be at ambient temperature, which is about 90°F (as reported by a thermometer on the wall).

The voltmeter reads about 17 mV, which is  $V_{\text{net}}$  in Equation 6.10:  $V_{\text{net}} = V_{\text{hot}} - V_{\text{cold}}$ . The graph of Figure 6.47 (for type J) is based on the cold junction being at freezing (32°F), which it certainly is not in this case. From the graph, we can see that the 90°F would create about 2.0 mV by itself. You can see

**Figure 6.48**

Measuring oven temperature with a thermocouple (Example 6.14).



from Equation 6.10 that if  $V_{\text{cold}}$  increases, it will reduce  $V_{\text{net}}$ , so if we are going to use the graph of Figure 6.47, we must compensate by increasing our reading of 17 mV to 19 mV.

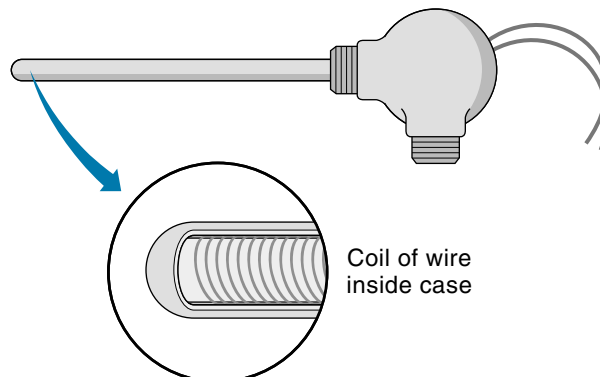
Now using the graph of Figure 6.47 for 19 mV, we read the temperature is 660°. This temperature is much lower than the desired 1000°F, so clearly there is something wrong with the temperature-control system.

## Resistance Temperature Detectors

The **resistance temperature detector** (RTD) is a temperature sensor based on the fact that metals increase in resistance as temperature rises. Figure 6.49 shows a typical RTD. A wire, such as platinum, is wrapped around a ceramic or glass rod (sometimes the wire coil is supported between two ceramic rods). Platinum wire has a temperature coefficient of  $0.0039 \Omega/\Omega/^\circ\text{C}$ , which means that the resistance goes up  $0.0039 \Omega$  for each ohm of wire for each Celsius degree of temperature rise. RTDs are available in different resistances, a common value being  $100 \Omega$ . Thus, a  $100\text{-}\Omega$  platinum RTD has a resistance of  $100 \Omega$  at  $0^\circ\text{C}$ , and it has a positive temperature coefficient of  $0.39 \Omega/^\circ\text{C}$ .

**Figure 6.49**

A resistance temperature detector (RTD).



**EXAMPLE 6.15**

A 100- $\Omega$  platinum RTD is being used in a system. The present resistance reading is 110  $\Omega$ . Find the temperature.

**SOLUTION**

A reading of 110  $\Omega$  means that the resistance has gone up 10  $\Omega$  from what it would be at 0°C. Therefore, knowing the temperature coefficient of RTD to be 0.39  $\Omega/^\circ\text{C}$ , we can calculate the current temperature:

$$10 \Omega \times \frac{^\circ\text{C}}{0.39 \Omega} = 25.6^\circ\text{C}$$

RTDs have the advantage of being very accurate and stable (characteristics do not change over time). The disadvantages are low sensitivity (small change in resistance per degree), relatively slow response time to temperature changes, and high cost.

**Thermistors**

A **thermistor** is a two-terminal device that changes resistance with temperature. Thermistors are made of oxide-based semiconductor materials and come in a variety of sizes and shapes. Thermistors are nonlinear; therefore, they are *not* usually used to get an accurate temperature reading but to indicate temperature changes, for example, overheating. Also, most thermistors have a negative temperature coefficient, which means the resistance decreases as temperature increases, as illustrated with the solid line in the graph of Figure 6.50(a). A very desirable feature of these devices is their high sensitivity. A relatively small change in temperature can produce a large change in resistance.

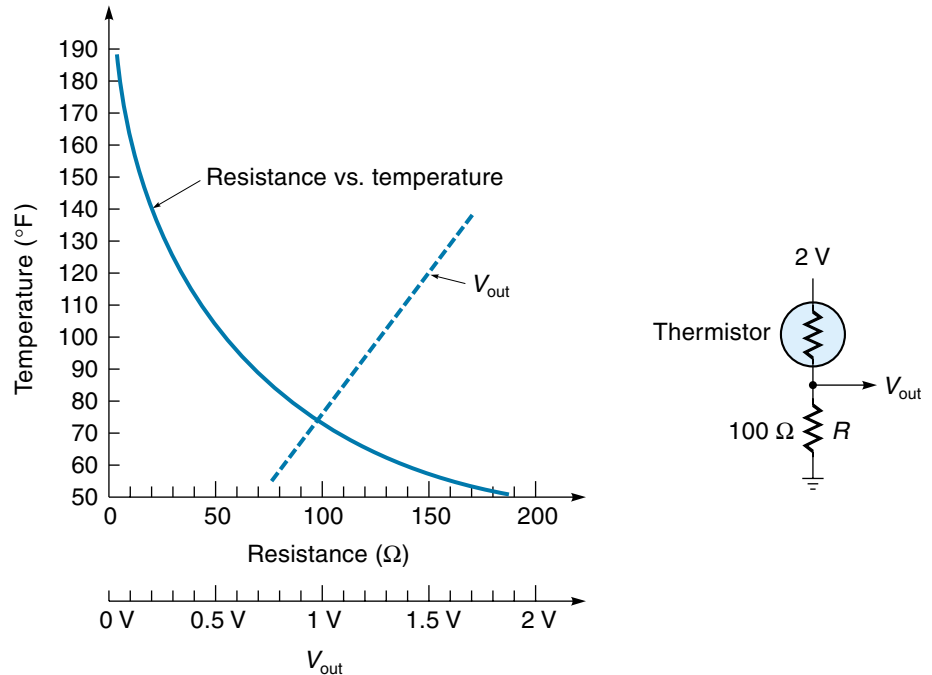
Figure 6.50(b) shows a simple thermistor interface circuit. By placing the thermistor in the top of a voltage divider, the resulting output voltage is relatively linear and has a positive slope [shown as a dashed line in Figure 6.50(a)]. The resistor ( $R$ ) value selected should be close to the nominal value of the thermistor.

Thermistors come in a wide range of resistances, from a few ohms to 1 M $\Omega$ , selection of which depends on the temperature range of interest. Higher-resistance models are used for higher temperatures, to increase the sensitivity, and to keep the sensor from drawing too much current. Consider, for example, what would happen if we used the thermistor of Figure 6.50 in the temperature range of 150-200°F; the sensitivity is only 0.1  $\Omega/^\circ\text{F}$ , and the nominal resistance is very low (15-20  $\Omega$ ). If we were to operate the same thermistor in the 50-100°F temperature range, the sensitivity is much higher (2.6  $\Omega/^\circ\text{F}$ ), and the nominal resistance is higher (between 50 and 180  $\Omega$ ).



**Figure 6.50**

A thermistor.



(a) Thermistor temperature vs. resistance curve

(b) Interface circuit

## Integrated-Circuit Temperature Sensors

Integrated-circuit temperature sensors come in various configurations. A common example is the LM34 and LM35 series. The LM34 produces an output voltage that is proportional to Fahrenheit temperature, and the LM35 produces an output that is proportional to Celsius temperature. Figure 6.51 gives the specification (spec) sheet for the LM35. Notice that it has three active terminals: supply voltage ( $V_s$ ), ground, and  $V_{out}$ .

The output voltage of the LM35 is directly proportional to °C, that is,

$$V_{out} = 10 \text{ mV}/^{\circ}\text{C}$$

This equation states that for each 1° increase in temperature, the output voltage increases by 10 mV. If only positive temperatures need to be measured, then the simple circuit shown in the spec sheet (bottom middle of Figure 6.51) can be used. If positive and negative temperatures must be measured, then the circuit on the bottom right can be used, which requires a positive- and negative-supply voltage.

**Figure 6.51**

The LM35 temperature sensor  
(Courtesy of National Semiconductor).



## LM35/LM35A/LM35C/LM35CA/LM35D Precision Centigrade Temperature Sensors

### General Description

The LM35 series are precision integrated-circuit temperature sensors, whose output voltage is linearly proportional to the Celsius (Centigrade) temperature. The LM35 thus has an advantage over linear temperature sensors calibrated in ° Kelvin, as the user is not required to subtract a large constant voltage from its output to obtain convenient Centigrade scaling. The LM35 does not require any external calibration or trimming to provide typical accuracies of  $\pm 1/4^\circ\text{C}$  at room temperature and  $\pm 3/4^\circ\text{C}$  over a full  $-55$  to  $+150^\circ\text{C}$  temperature range. Low cost is assured by trimming and calibration at the wafer level. The LM35's low output impedance, linear output, and precise inherent calibration make interfacing to readout or control circuitry especially easy. It can be used with single power supplies, or with plus and minus supplies. As it draws only  $60\ \mu\text{A}$  from its supply, it has very low self-heating, less than  $0.1^\circ\text{C}$  in still air. The LM35 is rated to operate over a  $-55^\circ\text{C}$  to  $+150^\circ\text{C}$  temperature range, while the LM35C is rated for a  $-40^\circ$  to  $+110^\circ\text{C}$  range ( $-10^\circ$  with improved accuracy). The LM35 series is

available packaged in hermetic TO-46 transistor packages, while the LM35C, LM35CA, and LM35D are also available in the plastic TO-92 transistor package. The LM35D is also available in an 8-lead surface mount small outline package and a plastic TO-202 package.

### Features

- Calibrated directly in ° Celsius (Centigrade)
- Linear + 10.0 mV/°C scale factor
- 0.5°C accuracy guaranteeable (at +25°C)
- Rated for full  $-55^\circ$  to  $+150^\circ\text{C}$  range
- Suitable for remote applications
- Low cost due to wafer-level trimming
- Operates from 4 to 30 volts
- Less than  $60\ \mu\text{A}$  current drain
- Low self-heating,  $0.08^\circ\text{C}$  in still air
- Nonlinearity only  $\pm 1/4^\circ\text{C}$  typical
- Low impedance output,  $0.1\ \Omega$  for 1 mA load

### Connection Diagrams

**TO-46  
Metal Can Package\***



BOTTOM VIEW

TL/H/5516-1

\*Case is connected to negative pin (GND)

Order Number LM35H, LM35AH,  
LM35CH, LM35CAH or LM35DH  
See NS Package Number H03H

**TO-92  
Plastic Package**

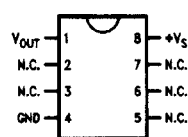


BOTTOM VIEW

TL/H/5516-2

Order Number LM35CZ,  
LM35CAZ or LM35DZ  
See NS Package Number Z03A

**SO-8  
Small Outline Molded Package**

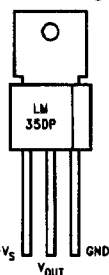


Top View

N.C. = No Connection

Order Number LM35DM  
See NS Package Number M08A

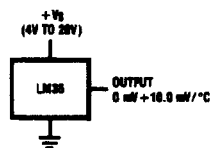
**TO-202  
Plastic Package**



TL/H/5516-24

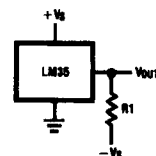
Order Number LM35DP  
See NS Package Number P03A

### Typical Applications



TL/H/5516-3

**FIGURE 1. Basic Centigrade  
Temperature  
Sensor (+2°C to +150°C)**



TL/H/5516-4

Choose  $R_1 = -V_S/50\ \mu\text{A}$

$V_{OUT} = +1,500\ \text{mV}$  at  $+150^\circ\text{C}$   
 $= +250\ \text{mV}$  at  $+25^\circ\text{C}$   
 $= -550\ \text{mV}$  at  $-55^\circ\text{C}$

**FIGURE 2. Full-Range Centigrade  
Temperature Sensor**

**EXAMPLE 6.16**

Construct a temperature sensor using the LM35 that has the following specifications:

Range: 5-100°C

Supply voltage: 5 V

Output: 0.1 V/°C

**SOLUTION**

The range requirement is no problem because the LM35 has an operating range of  $-55^{\circ}$  to  $150^{\circ}\text{C}$ . The task comes down to specifying the circuit and amplifying the output to meet the specifications.

Because the temperature range is positive, we can use the simple circuit designated as “Figure 1” in the spec sheet (Figure 6.51), using 5 V for the supply voltage.

The specifications also call for  $0.1\text{ V} = 1^{\circ}\text{C}$ , which is ten times greater than the LM35 output. This requirement can be met with the op-amp illustrated in Figure 6.52. (Op-amps are discussed in Chapter 3.) The gain of the amplifier can be set to 10 by proper selection of resistors:

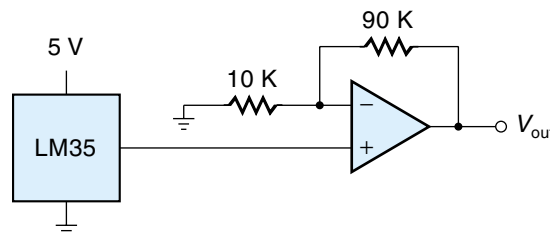
$$\text{Gain} = A = \frac{R_f}{R_a} + 1 = \frac{90\text{ k}\Omega}{10\text{ k}\Omega} + 1 = 10$$

The LM35 is a convenient IC to work with because the output voltage is in degrees Celsius. Some ICs, such as the LM135, provide an output that is in degrees kelvin. One degree of kelvin or Celsius represents the same interval of temperature, but the **Kelvin scale** starts at absolute zero temperature, which is  $273^{\circ}\text{C}$  below freezing. There is also an absolute zero temperature scale for Fahrenheit degrees, called the **Rankine scale**. These four temperature scales are compared in Figure 6.53.

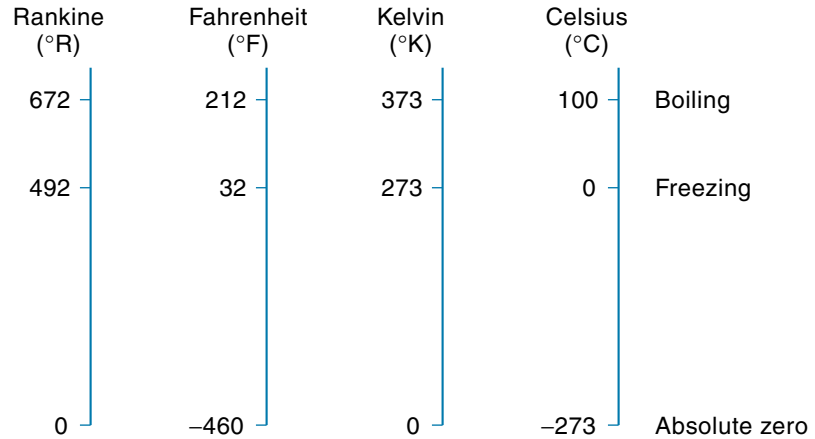
Another device, the AD7414 (Analog Devices), is a complete digital temperature-monitoring system. In a small, 6-pin IC, it has a temperature sensor, a 10-bit ADC, and a serial interface. It can also be programmed with high and low temperature limits; one of the output pins indicates when a programmed limit is exceeded.

**Figure 6.52**

An IC temperature sensor circuit (Example 6.16).



**Figure 6.53**  
Comparison of Rankine, Fahrenheit, Kelvin, and Celsius temperature scales.



Still another device, the TMP01 (made by Analog Devices), was designed specifically to be a single-chip thermostat. Three external resistors establish the upper and lower set points. The TMP01 outputs can directly drive relays to turn either heating or cooling on as needed.

## 6.7 FLOW SENSORS

**Flow sensors** measure the quantity of fluid material passing by a point in a certain time. Usually, the material is a gas or a liquid and is flowing in a pipe or open channel. The flowing of solid material, such as gravel traveling on a conveyor belt, will not be considered here. Flow transducers come in several types—those that use differential pressure, those where the flow spins a mechanical device, and a smaller class of sensors that use more sophisticated technologies.

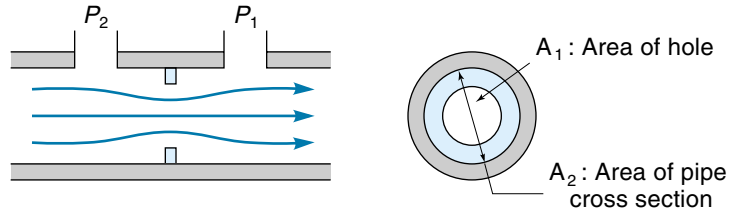
### Pressure-Based Flow Sensors

This group of flow sensors is based on the fact that pressure in a moving fluid is proportional to the flow. The pressure is detected with a pressure sensor; based on the physical dimensions of the system, the flow can be calculated. The simplest flow sensor is called the **orifice plate** (Figure 6.54), and is simply a restriction in the pipe that causes a pressure drop in the flow, much like a resistor that causes a drop in voltage in a circuit. This sensor requires two pressure ports, one upstream and one downstream of the restriction. The flow is proportional to the pressure difference between these ports and is calculated as follows:

$$Q = CA\sqrt{\frac{2g}{d}(P_2 - P_1)} \quad (6.11)$$

**Figure 6.54**

An orifice plate.



where

 $Q$  = flow (in<sup>3</sup>) $C$  = coefficient of discharge (approximately 0.63 for water if the orifice hole is at least half the pipe size) $A$  = area of the orifice hole (in<sup>2</sup>) $d$  = weight density of the fluid (lb/in<sup>3</sup>) $P_2 - P_1$  = pressure difference (psi) $g$  = gravity (384 in./s<sup>2</sup>)

The flow equation (6.11) is an approximation because, in addition to the pressure drop, the actual flow is dependent on velocity effects, the area ratio  $A_1/A_2$ , and the surface condition of the pipes. To get the correct constant ( $C$  in Equation 6.11) for a particular application, the flow sensor would have to be calibrated.

**EXAMPLE 6.17**

A flowmeter for water is made from a 3-in. ID (inside diameter) pipe. The orifice plate has a 2-in. opening. What is the flow if the pressure drop across the orifice plate is 0.2 psi?

**SOLUTION**

We will use Equation 6.11 to solve this problem, but we first need to figure out the values of the various terms:

$C = 0.63$  (we can start with this value because 2 in. is more than half of 3 in.)

$A = \pi r^2 = 3.14 \times (1)^2 = 3.14 \text{ in}^2$  (area of hole)

$g = 32 \text{ ft/s}^2 = 384 \text{ in./s}^2$  (gravity is a constant)

$P_2 - P_1 = 0.2 \text{ psi}$  (a given)

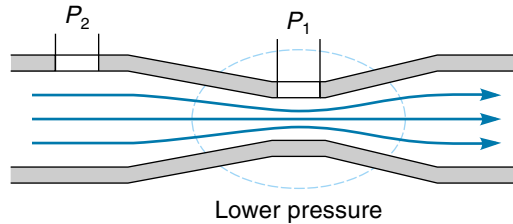
$d = 64.4 \text{ lb/ft}^3 = 0.037 \text{ lb/in}^3$  (weight density of water)

Now we can put these terms into Equation 6.11 and solve it:

$$Q = CA \sqrt{\frac{2g}{d}(P_2 - P_1)}$$

**Figure 6.55**

A venturi flow sensor.



$$= (0.63)(3.14 \text{ in}^2) \sqrt{\frac{2(384 \text{ in.})}{\text{s}^2} \frac{\text{in}^3}{0.037 \text{ lb}} \frac{(0.2 \text{ lb})}{\text{in}^2}} = 127 \text{ in}^3/\text{s}$$

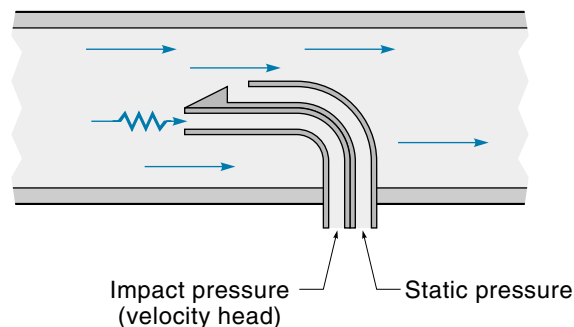
Therefore, 127 in<sup>3</sup>/s of water flow in the pipe. To make this a more accurate flowmeter, we would have to measure the flow with some other reliable flowmeter and then adjust  $C$  in the flow equation to make the readings consistent with what we know them to be.

Another pressure-based flow sensor uses the venturi to create the pressure difference, as illustrated in Figure 6.55. A **venturi** is a gradual restriction in the pipe that causes the fluid velocity to increase in the restricted area. This area of higher velocity has a lower pressure. The flow is proportional to the difference in pressure between  $P_2$  and  $P_1$ . The venturi flow sensor tends to keep the flow more laminar (smooth), but both the orifice plate and the venturi cause pressure drops in the pipe, which may be objectionable.

A pressure-based flow sensor that causes minimum restriction is the pitot tube. The **pitot tube** is a small open tube that faces into the flow (Figure 6.56). The probe actually consists of two tubes: One faces into the flow and reports the **impact pressure** (often called the *velocity head*), and one opens perpendicularly to the flow and reports the **static pressure**. The impact pressure is always greater than the static pressure, and

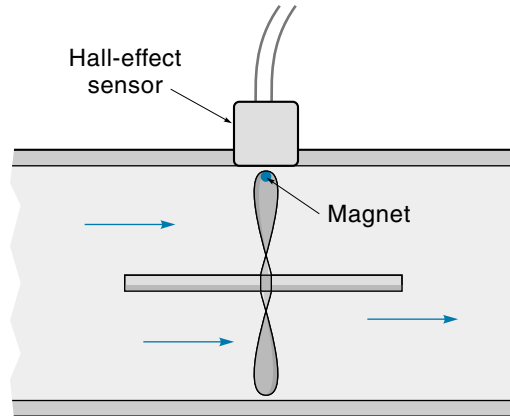
**Figure 6.56**

A pitot tube.



**Figure 6.57**

A turbine flow sensor.



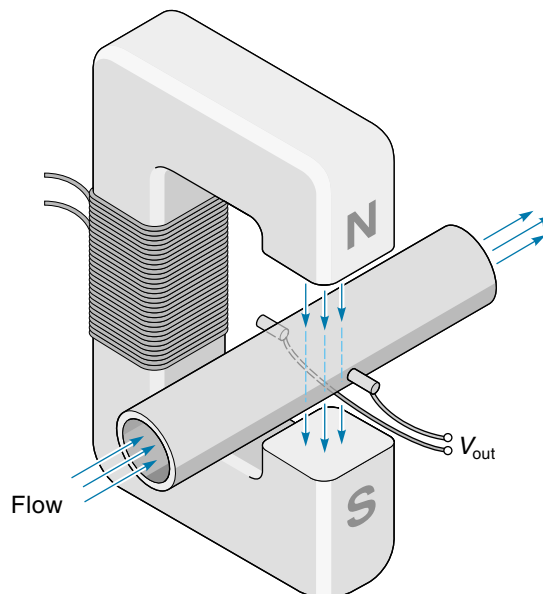
the difference between these two pressures is proportional to velocity and hence to flow. Common uses for the pitot tube are for aircraft and marine speed indicators.

### Turbine Flow Sensors

**Turbine**, or spin-type, flow sensors (also called flowmeters), employ a paddle wheel or propeller placed in the line of flow. The rotational velocity of the wheel is directly proportional to flow velocity. Figure 6.57 gives an illustration of this type of flow sensor. A small magnet is attached to one of the blades, and a Hall-effect sensor is mounted

**Figure 6.58**

A magnetic flowmeter.



in the housing. The Hall sensor gives a pulse for each revolution of the blades. The fact that the bearings are in the flow medium may eliminate this type of sensor for some applications, especially high-temperature or abrasive-type fluids.

### Magnetic Flowmeters

If a liquid is even slightly conductive (and many are), a magnetic flowmeter can be used. Shown in Figure 6.58, the magnetic flowmeter has no moving parts and presents no obstruction to the flow. A nonconducting section of pipe is placed in a magnetic field. The moving fluid in the pipe is like the moving conductor in a generator—it produces a voltage. The voltage, which is proportional to the fluid velocity, is detected from electrodes placed in the sides of the pipe.

---

## 6.8 LIQUID-LEVEL SENSORS

*Liquid-level sensors*, which measure the height of a liquid in a container, have two classifications: discrete and continuous. Discrete-level detectors can only detect whether the liquid is at a certain level. The continuous-level detector provides an analog signal that is proportional to the liquid level.

### Discrete-Level Detectors

**Discrete-level detectors** determine when a liquid has reached a certain level. An application of this type would be determining when to stop the fill cycle of a washing machine. The simplest type of level detector uses a float and a limit switch. There are many possible configurations of a float-based level detector—one is illustrated in Figure 6.59(a). In this case, the float is attached to a vertical rod. At a certain liquid level, the cam, which is attached to the rod, activates the limit switch. The activation level can be adjusted by relocating either the cam or the switch.

Another type of level detector is based on a photocell [Figure 6.59(b)]. When the liquid level submerges the light path, the photodetector signal changes, thus indicating the presence of the liquid.

Many liquids—such as tap water, weak acids, beer, and coffee (to name a few)—are slightly conductive, which offers another means of detection. As illustrated in Figure 6.59(c), an electric probe is suspended over the liquid. When the liquid reaches the probe, the resistance in the circuit abruptly decreases. A common use of this sensor is as an automotive low-coolant sensor.

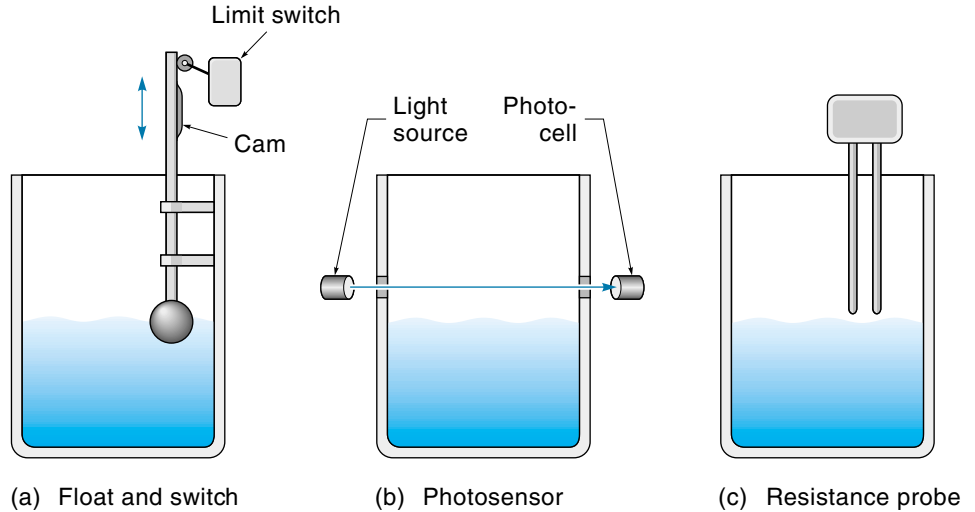
### Continuous-Level Detectors

**Continuous-level detectors** provide a signal that is proportional to the liquid level. There are a number of ways in which this can be done. One of the most direct methods



**Figure 6.59**

Discrete liquid-level detectors.



(used in the gas tank of your car) is a float connected to a position sensor. Figure 6.60(a) illustrates one implementation of this method.

Another way to measure liquid level is to measure the pressure at the bottom of the container [Figure 6.60(b)]. This method is based on the fact that the pressure at the bottom of the tank (called the **head**) is directly proportional to the level, as expressed in Equation 6.12:

$$\text{Pressure} = P = dH \quad (6.12)$$

where

$P$  = gauge pressure at the bottom (head)

$d$  = weight density (liquid weight per unit volume)

$H$  = height of liquid in the tank

### EXAMPLE 6.18

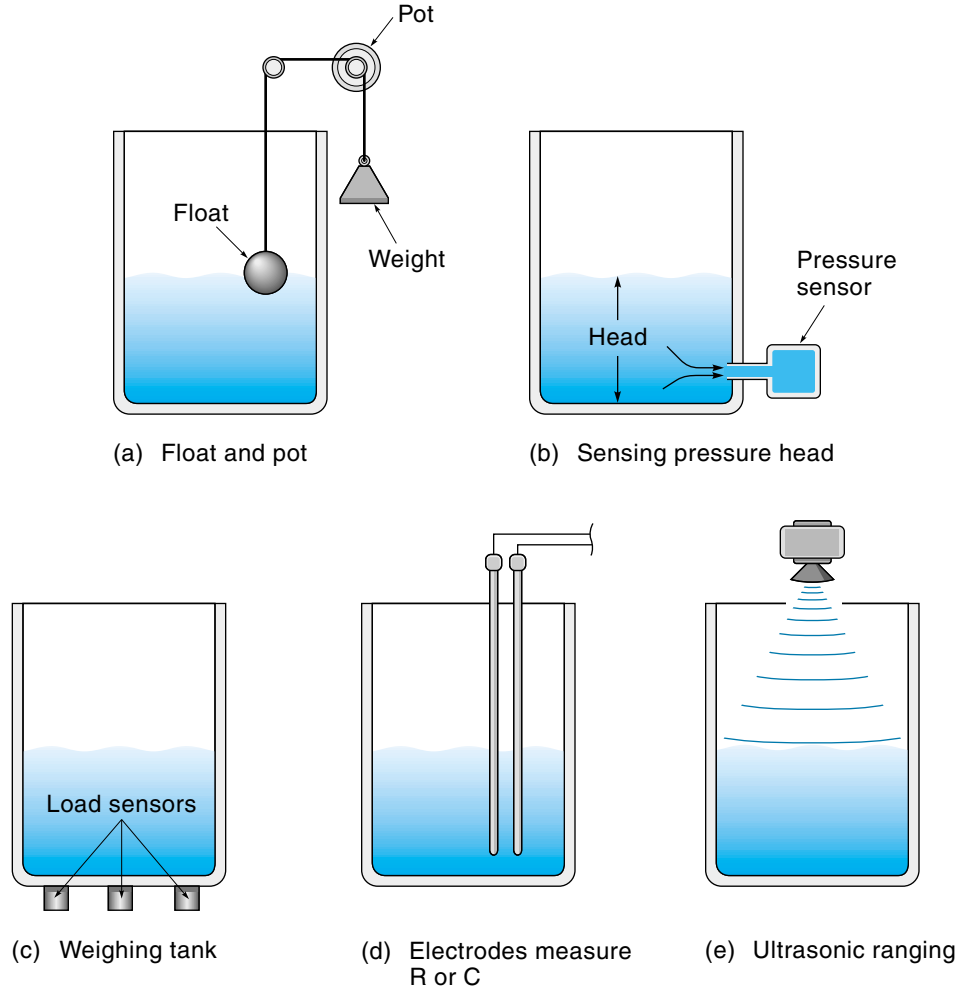
Calculate what the pressure (head) would be at the bottom of a 10-ft deep-water tank.

### SOLUTION

To use Equation 6.12, we need to know the weight density of the fluid (which can usually be found in a data book). The density for water is about 64 lb/ft<sup>3</sup>:

$$P = dH = 64 \text{ lb/ft}^3 \times 10 \text{ ft} = 640 \text{ lb/ft}^2$$

**Figure 6.60**  
Continuous-level  
detection methods.



We may want to convert this result to psi (lb/in<sup>2</sup>):

$$640 \text{ lb/ft}^2 \times \frac{1 \text{ ft}^2}{144 \text{ in}^2} = 4.44 \text{ lb/in}^2$$

Recall that 4.44 psi is the gauge pressure, which means the pressure at the bottom of the tank is 4.44 psi more than it is at the surface. The absolute pressure at the bottom would be

$$4.44 \text{ psi} + 14.7 \text{ psi} = 19.1 \text{ psi}$$

Gauge      Ambient      Absolute

**EXAMPLE 6.18 (Repeated with SI Units)**

Calculate what the pressure (head) would be at the bottom of a 3-m deep-water tank.

**SOLUTION**

To use Equation 6.12, we need to know the weight density of the fluid (which can usually be found in a data book). The weight density for water is  $9800 \text{ N/m}^3$ :

$$P = dH = \frac{9800 \text{ N}}{\text{m}^3} \times 3 \text{ m} = 29,400 \text{ N/m}^2 = 29,400 \text{ Pa}$$

The gauge pressure at the bottom of the tank is 29,400 Pa.

Monitoring the weight of the liquid with load cells is another technique that can determine liquid level [Figure 6.60(c)]. The level can then be calculated knowing the diameter and weight of the tank (empty) and the density of the fluid. Note that the total weight of the tank is the *sum* of the weights reported by the three load cells.

Some devices can detect the liquid level directly. The device shown in Figure 6.60(d) is simply two vertical electrodes mounted inside the tank. The output of the device, which must be amplified or otherwise processed, is either resistance or capacitance and is proportional to the level. Figure 6.60(e) shows another direct level-sensing system. This system uses an ultrasonic-range detector mounted over the tank. The complete unit, which includes the transducer and electronics, can be purchased as a module and is rather inexpensive.

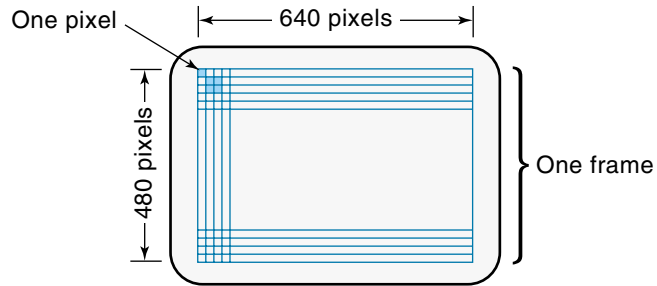
## 6.9 VISION SENSORS

A **vision sensor** is a TV camera connected to a computer. Machine vision is being used to perform inspections and to guide machine operations. For example, a system might use machine vision to determine whether parts had been made or assembled properly, or a vision system might be used to reject blemished oranges from a fruit-processing line. Alternatively, a vision system might be used to provide guidance to a pick-and-place robot for doing such things as unloading boxes from a pallet or inserting components in a circuit board.

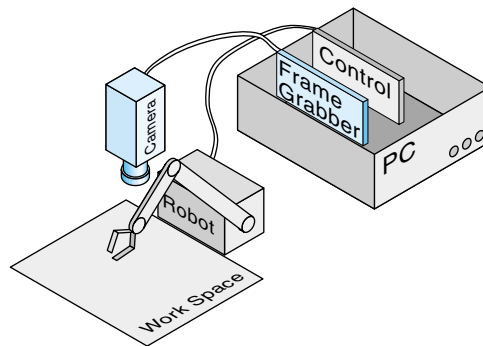
Vision systems require computing power to process thousands of pixels of information continuously in order to arrive at a go/no-go decision about what is “seen” [see Figure 6.61(a)]. In the past this could only be done with dedicated, expensive computing hardware. But two developments in PC technology have changed this situation. The first is the new generation of fast and powerful microprocessors such as those using Pentium with MMX technology. The second development is the adoption of the PCI

**Figure 6.61**

Vision system concepts.



(a) A frame is one “still picture” and is typically made up of 640 x 480 pixels



(b) Vision system components

expansion bus for PCs. PCI cards can communicate with the microprocessor at a much higher data rate than the older ISA cards could. Thus, technology has evolved to the point where relatively low-cost PCs have the power to process video information effectively, making vision systems practical. Vision-guided systems are now performing operations that before had to be done by hand or with very long and complicated but inflexible multi-sensor systems.

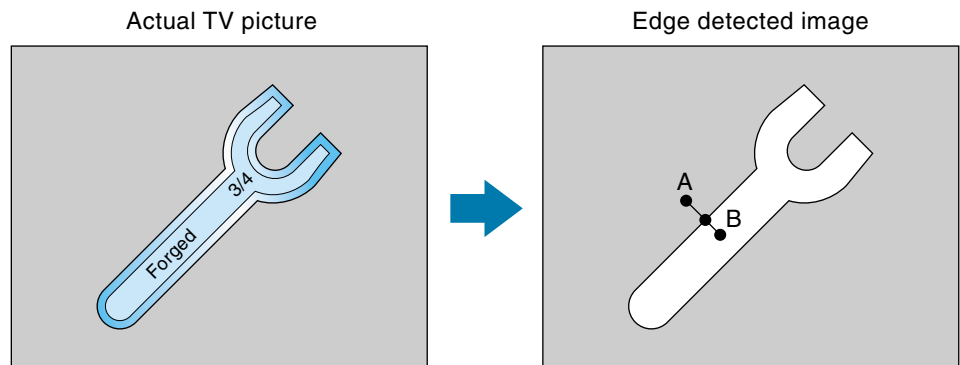
The hardware for a vision system is shown in Figure 6.61(b) and consists of a electronic camera connected to a special video-processing PCI card (called a *frame grabber*). The PCI card is plugged into a Pentium-based PC that is running special vision recognition software. Usually the camera is rigidly mounted so that it is looking down upon the area of interest. The visual information is passed from the camera to the frame grabber card. The card converts the visual information into digital format, which is typically a 640- by 480-pixel frame [see Figure 6.61(a)]. Each pixel is represented by an 8-bit gray-scale number called the *intensity*, meaning that each pixel is assigned one of 255 shades between black and white. Then this entire frame is passed on the PC (through the PC’s built-in PCI bus) up to 30 times a second. The PC is running soft-

ware that continuously analyzes the visual data the basis of some criterion and then sends signals back to the work area either to reject a part that has “failed” inspection or to direct a pick-and-place robot to move to a certain place to pick up a box.

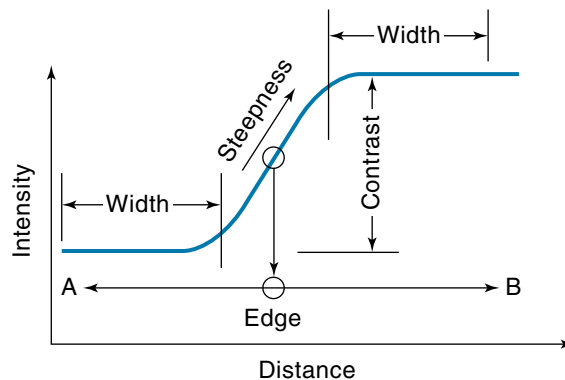
There are a number of different approaches to processing the visual data; which one used depends on the particular application. Probably the most common system is based on edge detection. In **edge detection**, the computer attempts to define the edges of the object. Figure 6.62(a) shows the results of the edge detection process. The computer scans the entire frame looking for edges—it detects them on the basis of changes in the gray-scale intensity. An edge is where there is a sharp intensity discontinuity. It is detected on the basis of three parameters: *contrast*, *width*, and *steepness*, as diagrammed in Figure 6.62b. Contrast is the numeric difference in intensity between two adjacent areas of the frame; steepness is how quickly the intensity changes between the areas; and width is the minimum size of the uniform-intensity area on either side of the edge. The actual edge is defined as the steepest place in the intensity curve. The threshold values of

**Figure 6.62**

Machine vision using edge detection.



(a) Computer uses edge detection to form an outline of the wrench



(b) Computer defines the edge on the basis of contrast, width, and steepness

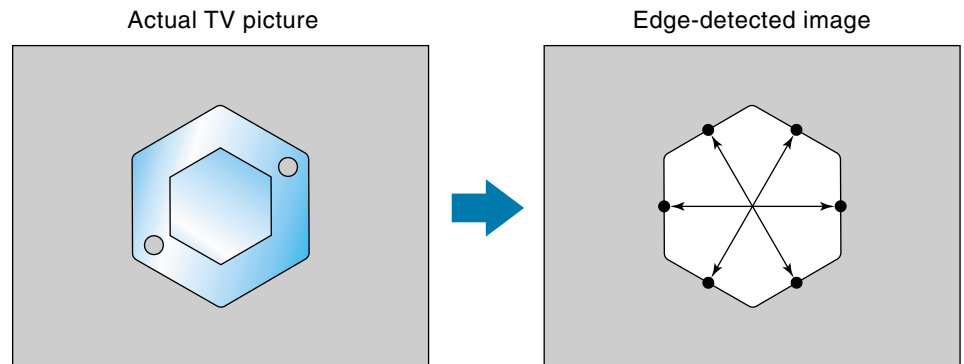
contrast, steepness, and width that the computer uses to try to identify edges can be tweaked to get the best performance in each particular application. Also, bright lighting and contrasting backgrounds can be used to accentuate the edges as much as possible.

Once the outline of an object is created within the computer, the dimensions can be checked on the visual image to determine whether the part is acceptable. For example, Figure 6.63 shows a system which inspects raw castings for completeness. Each hexagon shaped casting can be checked with three measurements. Also notice that the hexagons don't have to be in the same orientation on the assembly line. The software can be written to make the measurements with the casting in any orientation.

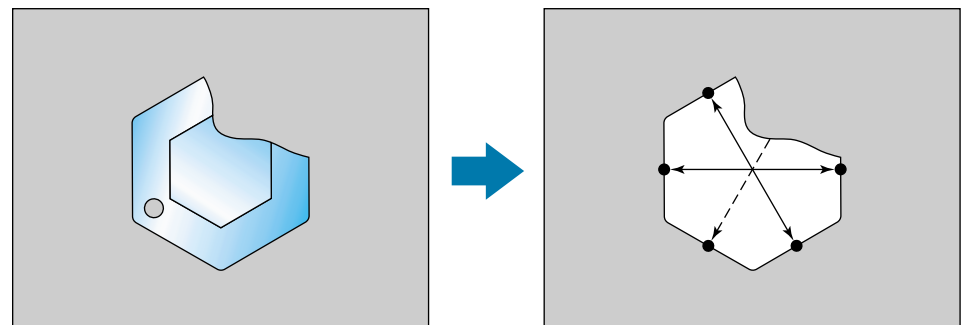
Another use of vision systems is to guide pick-and-place robots. For this task, the vision system tells the robot where the part is, and in what orientation, so that the robot can grab it. For example, consider the case where a robot needs to pick up the wrench in Figure 6.62(a) and put it in a box. Once the computer has generated the image on the basis of edge detection, it can process the shape mathematically to arrive at the centroid (middle) and the long axis. Using this information, the robot gripper can position itself over the wrench to pick it up.

**Figure 6.63**

Using a vision system to inspect hexagon castings.



(a) Computer locates six points and makes three dimension checks; the part is okay



(b) One dimension check fails; the part is rejected

Still another technique is called pattern recognition. *Pattern recognition* is used when an object needs to be identified. For example, in a system being used to sort parts into different bins, the computer maintains a library of the expected part shapes and, by using various “best-fit” algorithms, attempts to find the best match for each particular image shape. In another application, the job of the computer might be to scan the entire visual frame looking for a particular shape, such as a hole pattern, in order to guide a robot to mount a component.

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## SUMMARY

Sensors, also called transducers, are devices that sense physical parameters such as position, temperature, or pressure. In most cases, the sensor outputs an analog voltage (or digital value) that is proportional to the parameter being measured.

Position sensors measure the physical position of an object. Potentiometers (variable resistors) measure angular position and give an analog output. The optical encoder, another type of position sensor, uses a slotted disk and a photo sensor. The output of the optical encoder is in digital form. Linear motion can be detected with a linear-motion potentiometer or a linear variable differential transformer (LVDT). The LVDT uses a movable slug inside a special transformer. The phase and magnitude of the AC output can be processed to provide position information.

Position sensors can determine velocity by processing the data from two sequential position samples. The more direct method to measure velocity is to use a tachometer. A DC tachometer is actually a small generator that gives a DC voltage proportional to velocity. Optical tachometers give a pulse for each revolution of a shaft.

Proximity sensors sense whether an object has arrived at a certain place. The simplest way to do this is with a mechanical limit switch. Other methods would include using a photo sensor or a Hall-effect switch. The Hall-effect switch gives an output if a magnet is brought near a specially configured semiconductor.

Load sensors can determine force by measuring the small deformation that the force causes. The traditional method for measuring large forces is with a bonded-wire strain gauge. This device incorporates a pattern of thin wires. When stretched, the resistance of the wires change. Another method for measuring force uses the piezoresistive effect of semiconductors—that is, the resistance changes when the material is compressed.

Pressure sensors measure the pressure of liquids and gases. One class of pressure sensors, such as a bellows, uses the pressure to cause mechanical motion. Semiconductor pressure sensors convert pressure directly into electrical resistance.

A wide variety of temperature sensors are in use. Simple bimetallic strips will bend when heated and can then activate switch contacts. The thermocouple is a traditional high-temperature-sensing device that makes use of the fact that the junction of two dissimilar metals will create a small voltage when heated. The resistance temperature detector (RTD) uses the fact that a wire will increase in resistance when heated. Numerous semiconductor devices are available that “convert” temperature directly into resistance or voltage.

Flow sensors measure the flow of a fluid in a pipe or open channel. Many flow sensors work by placing a restriction in the pipe and then measuring the pressure before and after the restriction. The pressure difference between the two places is proportional to fluid velocity. Turbine flow sensors use the moving fluid to spin a propeller. The rpm of the propeller is proportional to velocity.

Liquid-level sensors determine the level of liquid in a tank. The discrete type can sense only if the level is at or above a certain point. Examples of the discrete level detectors are a float activating a limit switch or a photo sensor. The continuous-level detector gives an analog output proportional to fluid level. There are many different ways this can be done—for example, connecting a float to a potentiometer, monitoring the pressure at the bottom of the tank, or monitoring the weight of the tank with load cells.

Vision sensors are being used more and more for such things as inspecting parts and guiding robots in assembly operations. A vision system consists of a TV camera connected to a computer, typically a PC. Special vision software analyzes the video image and makes specific recommendations to the work area.

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## GLOSSARY

**absolute optical encoder** An optical rotary encoder that outputs a binary word representing the angular position.

**absolute pressure** The pressure difference between the measured value and an absolute vacuum.

**bimetallic temperature sensor** A strip made from two metals with different coefficients of thermal expansion; as the strip heats up, it bends.

**bonded-wire strain gauge** A force sensor consisting of a small pattern of thin wires attached to some structural member; when the member is stressed, the stretched wires slightly increase the resistance.

**Bourdon tube** A type of pressure sensor consisting of a short bent tube closed at one end; pressure inside the tube tends to straighten it.

**cold junction** One of two temperature-sensitive junctions of the thermocouple temperature sensor. The cold junction is usually kept at a reference temperature.

**compensating gauge** A nonactive strain gauge that is used solely for the purpose of canceling out temperature effects.

**continuous-level detector** A sensor that can determine the fluid level in a container.

**differential pressure** The difference between two pressures where neither may be ambient.

**direct current tachometer** Essentially, a DC generator that gives an output voltage proportional to angular velocity.



**discrete-level detector** A sensor that can determine if the fluid in a container has reached a certain level.

**edge detection** Used in machine vision systems; the process whereby a computer creates a visual image of an object by locating its edges.

**flow sensor** A sensor that measures the quantity of fluid flowing in a pipe or channel.

**gauge pressure** The difference between measured and ambient pressure (ambient is 14.7 psi or 101.3 kPa).

**Grey code** A sequence of digital states that has been designed so that only 1 bit changes between any two adjacent states.

**Hall effect** The phenomena of a semiconductor material generating a voltage when in the presence of a magnetic field; used primarily as a proximity sensor.

**head** The fluid pressure in a tank, which is caused by the weight of the fluid above and is therefore proportional to the level.

**Hooke's law** The deformation in a spring is directly proportional to the force on the spring (spring force = constant  $\times$  spring deformation).

**hot junction** One of two temperature-sensitive junctions of the thermocouple temperature sensor. The hot junction is used on the probe.

**ice-water bath** A traditional way to create a known reference temperature for the cold junction of a thermocouple.

**impact pressure** The pressure in an open (pitot) tube pointed “upstream.”

**incremental optical encoder** An optical rotary encoder that has one track of equally spaced slots; position is determined by counting the number of slots that pass by a photo sensor.

**Kelvin scale** The Kelvin temperature scale starts at absolute zero, but a Kelvin degree has the same temperature increment as a Celsius degree ( $0^{\circ}\text{C} = 273\text{ K}$ ).

**least significant bit (LSB)** The right-most bit in a binary number, it represents the smallest quantity that can be changed—that is, the difference between two successive states.

**limit switch** A switch used as a proximity sensor—that is, a switch mounted so that it is activated by some moving part.

**linearity error** A measurement error induced into the system by the sensor itself, it is the difference between the actual quantity and what the sensor reports it to be.

**linear variable differential transformer (LVDT)** A type of linear-motion position sensor. The motion of a magnetic core, which is allowed to slide inside a transformer, is proportional to the phase and magnitude of the output voltage.

**loading error** An error that may occur in an analog voltage signal when the circuit being driven draws too much current, thus “loading down” the voltage.

**LSB** *See* **least significant bit**.

**LVDT** *See* **linear variable differential transformer**.

**optical rotary encoder** A rotary position sensor that works by rotating a slotted disk past a photo sensor.

**optical tachometer** Mounted next to a rotating shaft, a photo sensor that gives a pulse for each revolution (a stripe is painted or fixed on the shaft).

**orifice plate** A type of flow sensor whereby a restriction is placed in a pipe causing a pressure difference (between either side of the restriction) that is proportional to flow.

**photodiode** A type of optical sensor, it increases its reverse-leakage current when exposed to light.

**photo resistor** A type of optical sensor; its resistance decreases when exposed to light.

**photo transistor** A type of optical sensor; light acts as the base current and turns on the transistor.

**photovoltaic cell** A device that converts light into electrical energy; used as an optical sensor, or as a solar cell.

**piezoresistive effect** A property of semiconductors in which the resistance changes when subjected to a force.

**potentiometer** A variable resistor that can be used as a position sensor.

**proximity sensor** A sensor that detects the physical presence of an object.

**psi** Pounds per square inch, a unit of pressure.

**pitot tube** A velocity sensor for fluids whereby a small open tube is placed directly into the flow; the pressure in the tube is proportional to fluid velocity.

**Rankine scale** The Rankine temperature scale starts at absolute zero, but a Rankine degree has the same temperature increment as a Fahrenheit degree ( $0^{\circ}\text{F} = 460^{\circ}\text{R}$ ).

**resistance temperature detector (RTD)** A temperature sensor based on the fact that the resistance of a metal wire will increase when the temperature rises.

**resolution** The smallest increment of data that can be detected or reported. For a digital system, the resolution is usually the value of the least significant bit.

**RTD** *See* **resistance temperature detector**.

**Seebeck effect** The property used by a thermocouple, a voltage proportional to temperature developed in a circuit consisting of junctions of dissimilar metal wires.

**sensor** A device that measures some physical parameter such as temperature, pressure, or position.

**slider (wiper)** The moving contact in a potentiometer, usually the center of three terminals.

**slotted coupler** An optical proximity sensor that is activated when an object “cuts” a light beam.

**static pressure** The pressure measured when the open measuring tube is directed perpendicularly to the flow.

**strain** The deformation (per unit length) as a result of stress.

**stress** Subjecting an object to tension or compression forces. Stress is the force per unit area within the object.

**thermistor** A temperature sensor based on the fact the resistance of some semiconductors will decrease as the temperature increases.

**thermocouple** A temperature-measuring sensor made from the junction of two dissimilar metal wires; when the junction is heated, a small voltage is generated.

**threshold detector** A circuit that provides a definite off-on signal when an analog voltage rises above a certain level.

**toothed-rotor tachometer** A tachometer that consists of two parts: a toothed, iron-based wheel attached to the rotating member, and a stationary sensor. The sensor gives a pulse each time a tooth passes by.

**transducer** A term used interchangeably with *sensor*. Literally means that energy is converted, which is what a sensor does.

**turbine** A flow sensor based on having the fluid rotate a propeller of some kind.

**variable-reluctance sensor** A magnet with a coil of wire around it. It can detect the movement of iron in the vicinity. Typically used with the toothed-rotor tachometer.

**venturi** A type of flow sensor whereby fluid is forced into a smaller channel, which increases its velocity; the higher velocity fluid has a lower pressure (than the fluid in the main channel), and the pressure difference is proportional to velocity.

**vision sensor** A system whereby a computer analyzes a video image for the purpose of inspecting parts or guiding a robot or machine.

**wiper** See **slider**.

**wire-wound potentiometer** A variable resistor that uses a coil of resistance wire for the resistive element; can be used as a position sensor.

**Young's modulus** A constant that relates stress and strain for a particular material (Young's modulus = stress/strain).

## EXERCISES

### Section 6.1

1. A potentiometer with a total range of  $350^\circ$  is supplied with a voltage of 12 Vdc. Find the output voltage for an angle of  $135^\circ$ .
2. A potentiometer with a total range of  $350^\circ$  is supplied with a voltage of 8 Vdc. The voltage at the wiper is 3.7 Vdc. What is the present angle of the pot?
3. A 10-k $\Omega$  pot is used as a position sensor with a 10-V supply. The input resistance to the interface circuit is 50 k $\Omega$ . Find the loading error when the pot is in the middle of its range.
4. A 5-k $\Omega$  pot is used as a position sensor with a 10-V supply. The input resistance to the interface circuit is 40 k $\Omega$ . Find the loading error when the pot is set to three-quarters of its range.
5. A potentiometer with a total range of  $350^\circ$  has a linearity of 0.25% and is connected to a 10-V source. If the pot is used as a position sensor, find the maximum possible error in degrees.
6. A  $350^\circ$  wire-wound potentiometer has 300 turns and a total resistance of 1 k $\Omega$ . What is the resolution in ohms? In degrees?
7. A  $350^\circ$  pot is connected to a shaft through a 3 : 1 gear ratio (the pot rotates three times further than the shaft). The pot has a linearity of 0.2%.
  - a. What is the maximum angle the shaft can turn?
  - b. What is the maximum difference in degrees from where the shaft is and where the pot reports it to be?
8. A pot with a total range of  $350^\circ$  is supplied with 5 Vdc. The output of the pot is converted to binary with an 8-bit ADC (which is also referenced at 5 Vdc). Find the 8-bit binary output of the ADC for a pot angle of  $60^\circ$ .
9. A  $350^\circ$  pot is connected through a 4 : 1 gear ratio to a shaft that rotates  $80^\circ$  (the pot rotates four times further than the shaft). The pot has a supply voltage of 5 V and feeds to an 8-bit ADC converter. The ADC has a reference voltage of 5 V. The LSB of the ADC is to be  $0.4^\circ$  or smaller. Will this system meet the requirements?
10. An absolute optical rotary encoder has five tracks. How many bits is the output, and what is the resolution in degrees per state (that is, degrees per LSB)?
11. An absolute optical rotary encoder in a certain application must have a resolution of  $3^\circ$ . How many tracks must it have?
12. An incremental optical rotary encoder has 720 slots. Starting from the reference point, the disk turns 200 slots CW, then 80 slots CCW, then 400 slots CW. What is the final angle of the shaft?
13. The counter for a 500-slot incremental optical rotary encoder has a value of 101100011.
  - a. What is the resolution in degrees—that is, what is the value of the LSB?
  - b. What is the current angle of the encoder shaft?

14. Compare an LVDT with a potentiometer as a position sensor. What are some advantages and disadvantages of using an LVDT?

### Section 6.2

15. Explain the basic principle of extracting velocity data from position sensors.
16. Position data from a pot is being processed to yield velocity data. The sample period is 0.5 s. The position is  $68^\circ$  at the first sample and  $73^\circ$  for the second sample. Calculate the velocity.
17. Data from an optical shaft encoder are providing velocity information. The sample period is 0.25 s. The present data are 10000111, and the previous data were 10000101. Find the present velocity ( $\text{LSB} = 1^\circ$ ).
18. A toothed-rotor sensor has 30 teeth. Find the rpm if the sensor outputs pulses at 100 Hz.
19. A toothed-rotor sensor has 18 teeth. Find the rpm if the sensor outputs pulses at 140 Hz.
20. The CK20-B tachometer is being used in a system. The output is 0.85 V. What is the velocity in rpm?
21. A CK20-B tachometer is being used in a system. The output is 1.5 V. What is the velocity in rpm?
22. A CK20-B tachometer is mounted piggyback to a motor with a 10:1 built-in gearbox (similar to Figure 6.27). Find the velocity of the output shaft when the tach outputs 2 V.

### Section 6.3

23. Describe one application of using a limit switch as a proximity sensor.
24. Both optical and Hall-effect sensors can be used as proximity switches. Name one possible application for each.
25. Explain the operation of a slotted coupler optical sensor. Use an application in your explanation.

### Section 6.4

26. When a 200-lb man gets in his car, the car body sinks 1 in. What is the spring constant for the suspension system of the car?
27. When a 180-lb man gets in his car, the car body sinks 1.25 in. What is the spring constant for the suspension system of the car?
28. When an 80-kg man gets in his car, the car body sinks 2 cm. What is the spring constant for the suspension system of the car? [*Hint*: Remember to convert the 80-kg mass into a weight using  $W = mg$ .]
29. How does a bonded-wire strain gauge work as a force sensor?
30. What is the purpose of using a bridge circuit with a strain gauge, and what is the purpose of the compensating gauge?

31. A strain gauge is used to measure the tension force in a 1-in. diameter bar of steel. The strain gauge has a nominal resistance of  $140\ \Omega$  and a gauge factor (GF) of 4. The strain gauge is connected to a bridge (which is supplied with 10 Vdc). The bridge was initially balanced. After the bar is put under tension, the bridge output voltage goes to 0.0008 Vdc. Draw a schematic of the setup and calculate the force on the bar.
32. A strain gauge is used to measure the tension force in a 0.75-in. diameter bar of steel. The strain gauge has a nominal resistance of  $180\ \Omega$  and a GF of 4. The strain gauge is connected to a bridge (which is supplied with 10 Vdc). The bridge was initially balanced. After the bar is put under tension, the bridge output voltage goes to 0.0006 Vdc. Draw a schematic of the setup and calculate the force on the bar.
33. A strain gauge is used to measure the tension force in a 2-cm diameter bar of steel. The strain gauge has a nominal resistance of  $160\ \Omega$  and a GF of 5. The strain gauge is connected to a bridge (which is supplied with 10 Vdc). The bridge was initially balanced. After the bar is put under tension, the bridge output voltage goes to 0.0008 Vdc. Draw a schematic of the setup and calculate the force on the bar.
34. A low-force sensor uses a spring and a  $1\text{-k}\Omega$  linear pot. The pot has a 1-in. stroke and a supply voltage of 10 Vdc. If the sensor is to have an output of 1 V per 10 lb of load, what is the spring constant required?

### Section 6.5

35. How does a Bourdon tube work as a pressure sensor?
36. What is the difference between *gauge pressure*, *differential pressure*, and *absolute pressure*?
37. A differential pressure bellows sensor is receiving 4 psi (gauge) on one side and 12 psi (absolute) on the other. What pressure does the sensor report?
38. A differential pressure bellows sensor is receiving 5 psi (gauge) on one side and 16.2 psi (absolute) on the other. What pressure does the sensor report?
39. A differential pressure bellows sensor is receiving 20 kPa (gauge) on one side and 90 kPa (absolute) on the other. What pressure does the sensor report?

### Section 6.6

40. How does a bimetallic temperature sensor work?
41. List the three ways that the cold junction of a thermocouple is handled in modern systems.
42. An iron-constantan thermocouple is referenced at  $32^\circ\text{F}$  and has an output voltage of 45 mV. What is the temperature at the hot junction?
43. An iron-constantan thermocouple is used to measure the temperature in an oven. The cold junction is at ambient temperature of  $80^\circ\text{F}$ , and the thermocouple voltage is 12 mV. What is the approximate oven temperature?
44. A  $100\text{-}\Omega$  platinum RTD is being used to measure temperature in an oven. The present resistance is  $122\ \Omega$ . Find the temperature.

45. A 100- $\Omega$  platinum RTD is being used to measure temperature in an oven. The present resistance is 130  $\Omega$ . Find the temperature.
46. Construct a temperature sensor using the LM35 with the following specifications: Supply voltage = 12 V, and output = 0.5 V/ $^{\circ}$ C.

### Section 6.7

47. Describe the operating principles of the following flow sensors:
  - a. Orifice plate
  - b. Venturi
  - c. Pitot tube
  - d. Turbine
  - e. Magnetic
48. An orifice plate flow sensor is used to measure the flow of water in a 2-in. ID pipe. The orifice plate has a 1.5-in. diameter hole. The pressure difference across the plate is 0.3 psi. Find the approximate water flow in the pipe.

### Section 6.8

49. How would you use two discrete-level detectors to maintain the liquid level in a tank at between 3 and 4 ft?
50. A pressure sensor is used to measure the water level in a tank. What pressure would you expect for 5 ft of water?
51. A pressure sensor is used to measure the water level in a tank. What pressure would you expect for 2 m of water?
52. A pressure sensor is used to measure the level of a liquid in a tank. The density of the liquid is 52 lb/ft<sup>3</sup>. If the gauge pressure (from the sensor) is 5.7 psi, what is the height of the liquid?
53. A pressure sensor is used to measure the level of a liquid in a tank. The weight density of the liquid is 6000 N/m<sup>3</sup>. If the gauge pressure (from the sensor) is 12,000 Pa, what is the height of the liquid?

### Section 6.9

54. Describe the basic components of a machine vision system.
55. What process would a machine vision system go through to determine whether a part was incorrectly made or assembled?





# CHAPTER 7

## Direct Current Motors

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### OBJECTIVES

After studying this chapter, you should be able to:

- Explain the theory of operation of electric motors in general and DC motors in particular.
- Distinguish the characteristics of series-wound, shunt-wound, compound, and permanent-magnet motors.
- Use the torque-speed curve of a motor to predict its performance.
- Select a DC motor based on mechanical requirements.
- Understand the operation of linear amplifier drivers for DC motors that incorporate power transistors, IC amplifiers, Darlington transistors, and power MOSFETS.
- Understand DC motor-speed control using pulse-width-modulation concepts.
- Understand operating a DC motor from rectified AC, using silicon-controlled-rectifier circuits.
- Understand the operating principles of brushless DC motors.

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### INTRODUCTION

An indispensable component of the control system is the actuator. The actuator is the first system component to actually move, converting electrical energy into mechanical motion. The most common type of actuator is the electric motor.

Motors are classified as either DC or AC, depending on the type of power they use. AC motors (covered in Chapter 9) have some advantages over DC motors: They tend to be smaller, more reliable, and less expensive. However, they generally run at a fixed speed that is determined by the line frequency. DC motors have speed-control capability, which means that speed, torque, and even direction of rotation can be changed at any time to meet new conditions. Also, smaller DC motors commonly operate at lower voltages (for example, a 12-V disk drive motor), which makes them easier to interface with control electronics.

## 7.1 THEORY OF OPERATION

The discovery that led to the invention of the electric motor was simply this: A current-carrying conductor will experience a force when placed in a magnetic field. The conductor can be any metal—iron, copper, aluminum, and so on. The direction of the force is perpendicular to both the magnetic field and the current (Figure 7.1). A demonstration of this principle is easy to perform with a strong magnet, flashlight battery, and a wire and is highly recommended! Place the wire between the magnet poles and alternately connect and disconnect the wire from the battery. Each time you complete the circuit, you should feel a little tug on the wire. The magnitude of the force on the wire can be calculated from the following equation:

$$F = IBL \sin \theta \quad (7.1)$$

where

$F$  = force on the conductor (in Newtons)

$I$  = current through the conductor (in amperes)

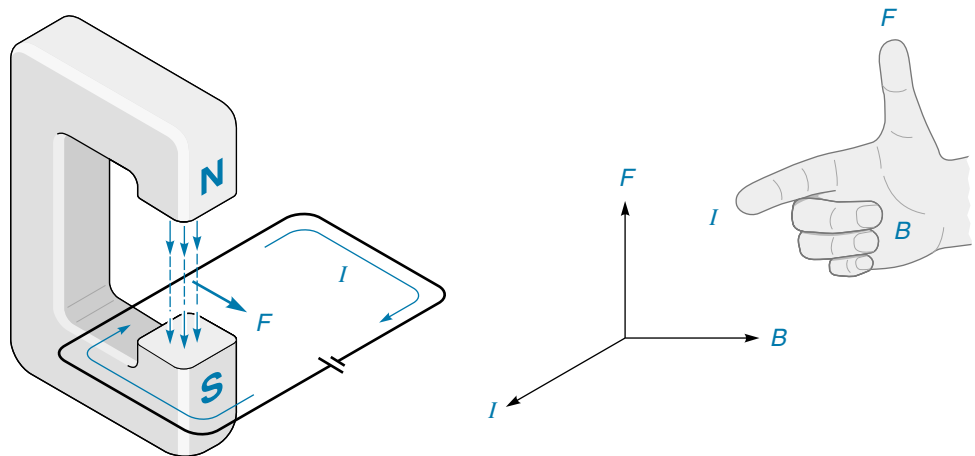
$B$  = magnetic flux density (in gauss)

$L$  = length of the wire (in meters)

$\theta$  = angle between the magnetic field and current

An electric motor must harness this force in such a way as to cause a rotary motion. This can be done by forming the wire in a loop and placing it in the magnetic field (Figure 7.2). The loop (or *coil*) of wire is allowed to rotate about the axis shown and is

**Figure 7.1**  
Action of force on a wire in a magnetic field.

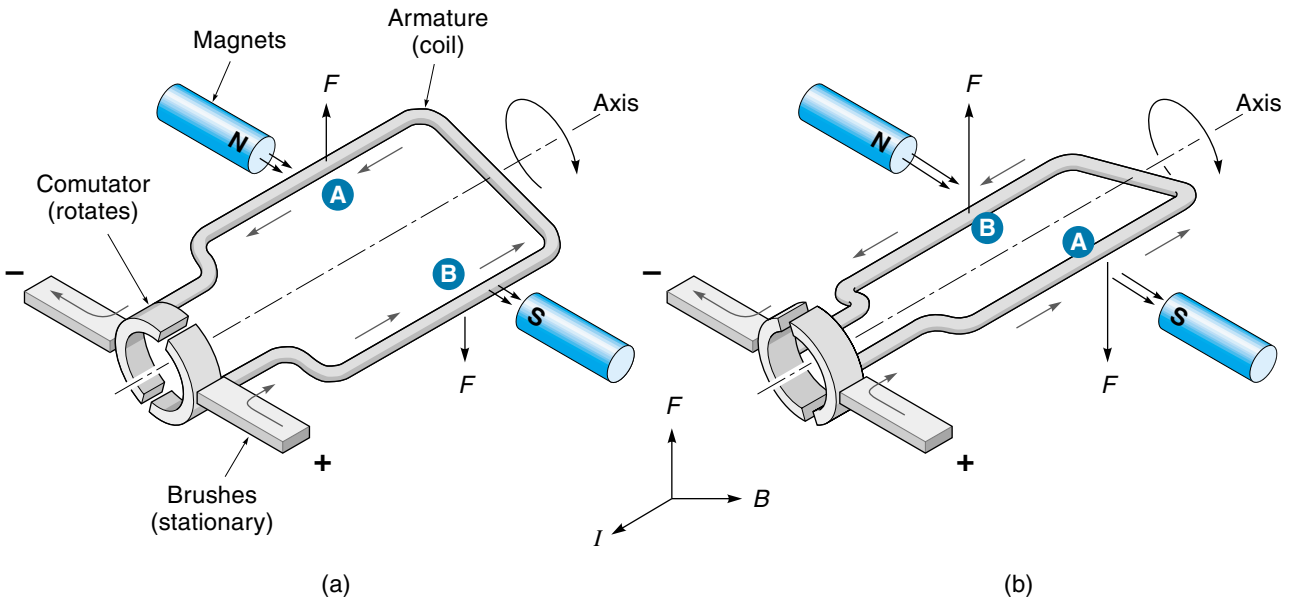


(a) Experimental setup

(b) Directions of  $I$ ,  $F$ , and  $B$  are mutually orthogonal (conventional current flow)

**Figure 7.2**

A simple DC motor action  
(conventional flow).



called the **armature** winding. The armature is placed in a magnetic field called the **field**. The **commutator** and **brushes** supply current to the armature while allowing it to rotate.

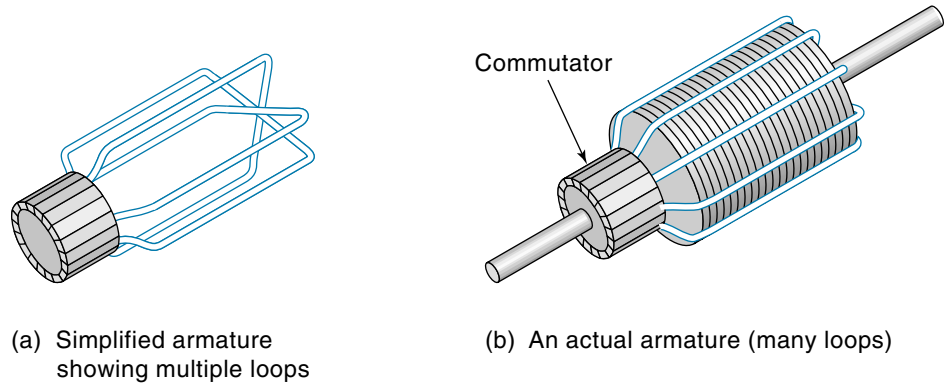
To understand how the motor works, look at Figure 7.2(a). Notice that wire segments A and B of the coil are in the same magnetic field, but the current in wire segment A is coming out of the page, whereas the current in wire segment B is going in. Applying the force diagram from Figure 7.1(b) (and repeated in Figure 7.2), we see that wire segment A of the coil would be forced up, whereas wire segment B would be forced down. These forces would cause the coil to rotate clockwise. Figure 7.2(b) shows the situation after the coil has rotated about  $90^\circ$ . The current has now reversed direction in the coil because the commutator contacts have rotated and are now making contact with the opposite brush. Now wire segment A of the coil will be forced down and wire segment B up, which causes the armature to continue rotating clockwise. **Torque**, as explained in Chapter 5, is the rotational force a motor can exert. Notice that the torque will be at a maximum when the coil is horizontal and will drop to zero when the coil is vertical (similar to peddling a bicycle).

To reverse the direction of the motor, polarity of the voltage to the commutator is reversed. This causes the forces on the armature coil to be reversed, and the motor would then run in the opposite direction.

Figure 7.3 shows the armature of a practical motor. Notice that there are multiple coils and each coil experiences the forces described in the preceding paragraph and so contributes to the overall torque of the motor. Each coil is connected to a separate pair

**Figure 7.3**

DC motor armature.



of commutator segments, causing the current in each coil to switch directions at the proper time for that individual coil. The overall effect is to provide approximately the same torque for any armature position (like a multipiston engine).

One of the most important operating parameters of any motor is torque. Electric motor torque is directly proportional to the force on the armature wires. From Equation 7.1, we see that the force is proportional to the magnetic field and current (not voltage). By gathering the mechanical parameters of the motor (such as number of poles) into a single constant  $K$ , the motor torque can be expressed as

$$T = K_T I_A \phi \quad (7.2)$$

where

$T$  = motor torque

$K_T$  = a constant based on the motor construction

$I_A$  = armature current

$\phi$  = magnetic flux

So far we have been looking at how the motor converts electrical energy to mechanical energy. It turns out that the very same device (motor) is also capable of converting mechanical energy to electrical energy, in which case it is called a generator. For example, if the armature coil of Figure 7.2 were rotated in the magnetic field by some external force, a voltage [called the *electromotive force* (EMF)] would appear on the commutator segments. The magnitude of the EMF is given in Equation 7.3:

$$\text{EMF} = K_E \phi S \quad (7.3)$$

where

EMF = voltage generated by the turning motor

$K_E$  = a constant based on motor construction

$\phi$  = magnetic flux

$S$  = speed of motor (rpm)

Although it may seem strange, this EMF voltage is being generated even when the motor is running on its own power, but it has the opposite polarity of the line voltage; hence, it is called the **counter-EMF** (CEMF). *Its effect is to cancel out some of the line voltage.* In other words, the actual voltage available to the armature is the line voltage minus the CEMF:

$$V_A = V_{\text{in}} - \text{CEMF} \quad (7.4)$$

where

$V_A$  = actual voltage available to the armature

$V_{\text{in}}$  = line voltage supplied to the motor

CEMF = voltage generated within the motor

You can not directly measure  $V_A$  with a voltmeter because it is an effective voltage inside the armature. However, there *is* physical evidence that the CEMF exists because the armature current is also reduced, as indicated in Equation 7.5:

$$I_A = \frac{V_{\text{in}} - \text{CEMF}}{R_A} \quad (7.5)$$

where

$I_A$  = armature current

$V_{\text{in}}$  = line voltage to the motor

$R_A$  = armature resistance

CEMF = voltage generated within the motor

Equation 7.5 (which is in the form of Ohm's law) tells us that the armature current is a function of the applied voltage minus the CEMF. Because CEMF increases with motor speed, the faster the motor runs, the less current the motor will draw, and consequently its torque will diminish. This explains why most DC motors have a finite maximum speed; eventually, if the motor keeps going faster, the CEMF will nearly cancel out the line voltage, and the armature current will approach zero.

The actual relationship between motor speed and CEMF follows and is derived from Equation 7.3:

$$S = \frac{\text{CEMF}}{K_E \phi} \quad (7.6)$$

where

$S$  = speed of the motor (rpm)

CEMF = voltage generated within the motor

$K_E$  = a motor constant

$\phi$  = magnetic flux

Looking at Equation 7.6, we see that the motor speed is directly proportional to the CEMF (voltage) and (surprisingly) inversely proportional to the field flux.

**EXAMPLE 7.1**

A 12 Vdc motor has an armature resistance of  $10\ \Omega$  and according to its spec sheet generates a CEMF at the rate of  $0.3\ \text{V}/100\ \text{rpm}$ . Find the actual armature current at 0 rpm and at 1000 rpm.

**SOLUTION**

We can find the armature current with Equation 7.5. For the first case, when the motor isn't turning at all (0 rpm), the CEMF will be 0 V:

$$I_A = \frac{V_{\text{in}} - \text{CEMF}}{R} = \frac{12\ \text{V} - 0\ \text{V}}{10} = 1.2\ \text{A}$$

For the second case (1000 rpm), determine the CEMF before applying Equation 7.5. Given the CEMF rate of  $0.3\ \text{V}/100\ \text{rpm}$ ,

$$\text{CEMF} = \frac{0.3\ \text{V}}{100\ \text{rpm}} \times 1000\ \text{rpm} = 3\ \text{V}$$

Then

$$I_A = \frac{V_{\text{in}} - \text{CEMF}}{R_a} = \frac{12\ \text{V} - 3\ \text{V}}{10} = \frac{9\ \text{V}}{10} = 0.9\ \text{A}$$

Thus, we see that the armature current is reduced in the running motor.

DC motors have a property called speed regulation. **Speed regulation** is the ability of a motor to maintain its speed when the load is applied. The basis of this self-regulation is the CEMF. When the motor's load is increased, the speed tends to decrease, but the lower speed reduces the CEMF, which allows more current into the armature. The increased current results in increased torque, which prevents the motor from slowing further. Speed regulation is usually expressed as a percentage, as shown in Equation 7.7:

$$\% \text{ speed regulation} = \frac{S_{\text{NL}} - S_{\text{FL}}}{S_{\text{FL}}} \times 100 \quad (7.7)$$

where

$S_{\text{NL}}$  = no-load speeds

$S_{\text{FL}}$  = full-load speeds

**EXAMPLE 7.2**

A motor has a no-load speed of 1150 rpm. When the maximum load for a certain application is applied to the motor, the speed drops to 1000 rpm. Find the speed regulation for this application.

**SOLUTION**

Applying Equation 7.7, we get

$$\frac{S_{NL} - S_{FL}}{S_{FL}} \times 100 = \frac{1150 \text{ rpm} - 1000 \text{ rpm}}{1000 \text{ rpm}} \times 100 = 15.0\%$$

**7.2 WOUND-FIELD DC MOTORS**

Wound-field motors use an electromagnet called the **field winding** to generate the magnetic field. The only other way to generate a magnetic field is with permanent magnets, which will be discussed later. The speed of wound-field motors is controlled by varying the voltage to the armature or field windings. Figure 7.4 shows an example of a

**Figure 7.4**

Wound-field DC motor. (Explosion-proof means no open sparks.)  
(Courtesy of Reliance Electric)

**FEATURES**

- ADJUSTABLE SPEED
- 90 V ARMATURE 50/100 V FIELD(3)
- CLASS F INSULATION
- 40°C AMBIENT
- EXPLOSION-PROOF(1)
- FOR OPERATING FROM FULL WAVE SINGLE-PHASE RECTIFIED POWER (TYPE K)
- RIGID MOUNT C-FACE



**WOUND FIELD**  
**EXPLOSION-PROOF**  
**1/4-1 HP**  
**TOTALLY ENCLOSED • FAN-COOLED**

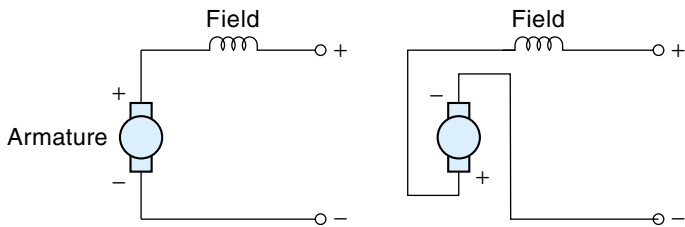
HP	RPM	Enclosure	Frame #
1/4	3500	TEFC-XP	HM56HC
	2500	TEFC-XP	HB56HC
	1750	TEFC-XP	HB56HC ♦
	1150	TEFC-XP	HU56HC
1/3	3500	TEFC-XP	HG56HC
	2500	TEFC-XP	HU56HC
	1750	TEFC-XP	HB56HC ♦
	1150	TEFC-XP	HJ56HC
1/2	3500	TEFC-XP	HU56HC
	2500	TEFC-XP	HG56HC
	1750	TEFC-XP	HJ56HC ♦
	1150	TEFC-XP	HE56HC
3/4	3500	TEFC-XP	HU56HC
	2500	TEFC-XP	HJ56HC
	1750	TEFC-XP	HJ56HC ♦
1	3500(2)	TEFC-XP	HJ56HC
	2500(2)	TEFC-XP	HE56HC
	1750(2)	TEFC-XP	HE56HC ♦

wound-field motor. This particular series of motors is available from  $\frac{1}{4}$  to 1 hp and at several standard rated speeds. The **rated speed** of a motor is the speed<sup>4</sup> when it is supplying the rated horsepower—when the motor is unloaded, it will go faster than the rated speed. These motors are designed to run at 90 Vdc because 90 V is about what a practical rectifier circuit can produce from standard 120 Vac. The speed can be controlled by adjusting the rectified voltage, as will be discussed later in this chapter. There are three basic types of wound field motors: series wound, shunt wound, and compound.

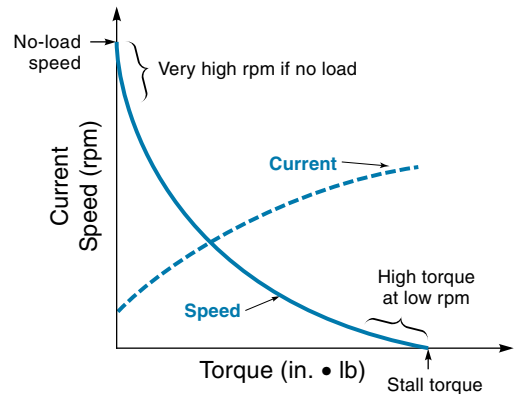
### Series-Wound Motors

In a **series-wound motor**, the armature and field windings are connected in series [Figure 7.5(a)]. This configuration gives the motor a large starting torque, which is useful in many situations—for example, car starter motors. The explanation for this large

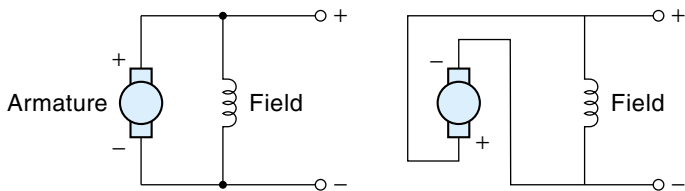
**Figure 7.5**  
Series and shunt  
DC motors.



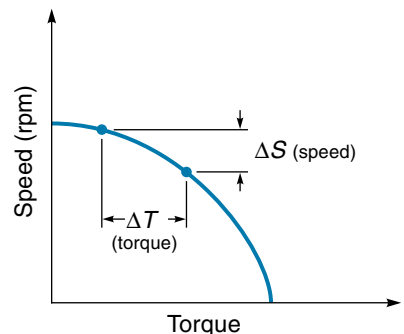
(a) Series motor circuit (b) Series motor reversed



(c) Torque–speed curve for series motor



(d) Shunt motor circuit (e) Shunt motor reversed



(f) Torque–speed curve for shunt motor



initial torque is as follows: When the motor is stopped, there is no CEMF, and the full-line voltage is available to the windings. Therefore, the initial armature current is large; and being in series with the field windings, this same current creates a large magnetic field as well. The combination of a large armature current *and* a large field flux is what produces the large start-up torque (see Equation 7.2). Once the motor starts turning, the increasing CEMF reduces the motor currents and hence the torque. Because the field coil carries the full armature current, it must have a low resistance; thus, it consists of a few turns of heavy-gauge wire.

Another characteristic of the series-wound motor is that it tends to “run away” (go faster and faster) under no-load conditions. As the field current diminishes from the increasing CEMF, the magnetic field flux also decreases, which according to Equation 7.6 tends to speed up the motor, which increases the CEMF even more. The overall effect of this seemingly circular logic is that the motor will continue to accelerate until the torque is balanced by friction forces. Most smaller motors can handle the high speeds without causing any damage, but larger motors may literally fly apart if operated with no load.

Figure 7.5(c) shows a typical **torque-speed curve** for a series-wound motor. Understanding this graph is important because it is the most useful tool in describing a particular motor’s operating characteristics. The vertical axis has two scales: speed and current. The horizontal axis represents torque. As described earlier, torque is a measure of the motor’s strength in turning its shaft and is the direct result of the forces on the armature conductors. The graph shows the torque-speed relationship for a particular constant applied voltage. A lower motor voltage would give a curve of the same shape but lying to the left of the one shown.

Looking at the curve [Figure 7.5(c)], notice that *speed decreases as torque increases*. This is true of all electric motors, and it makes intuitive sense; when the load is increased, the motor slows down. As an example, consider an electric circular saw. You can hear the motor slowing down under the load of each board being cut. *The maximum torque the DC motor can deliver occurs when the motor is loaded so much it comes to a stop*. This is called the **stall torque**. Some motors are designed to operate right down to the stall torque; in fact, many control system motors operate in this mode all the time. For example, a motor driving a robot arm must move the loaded arm from one rest position to another. A useful coincidence is that the stall torque is also the maximum torque because most mechanical systems require more force to get things moving from a resting position than they do at other times.

The top end of the curve [Figure 7.5(c)] reveals the **no-load speed**. This is the speed the motor would attain if it were allowed to run with absolutely no external load on it. In series motors, the no-load speed is established by a number of factors, including bearing friction and wind resistance to the rotating parts, and so is somewhat unpredictable. In other types of motors, the effect of CEMF is the primary factor in determining the no-load speed. Notice that at the no-load speed the motor is capable of doing absolutely no useful work because it is supplying zero torque to the outside world. Therefore, *any motor that is doing something useful must be going at*

*less than its no-load speed.* As pointed out before, larger series motors could be damaged if run with no external load and, for that reason, are never connected to a load with a belt—belts break!

The other parameter graphed in Figure 7.5(c) is the current (shown as a dashed line). The current increases as the torque increases because the force on the armature conductors is proportional to armature current. It may seem backward that when the motor turns faster it draws less current, but a faster turning motor means it has less of a load on it.

The speed-torque curve of the series-wound motor is highly nonlinear. The unusually high-stall torque is a desirable quality for many workhorse applications such as cranes, portable power tools, and automobile starter motors. Series motors are less desirable in control applications, however, because the nonlinear characteristics complicate the mathematics done by the controller. One last comment on the series motor: Like all wound field motors, reversing the polarity of the line voltage reverses both the armature and field windings, *causing the motor to continue to rotate in the same direction.* To reverse the direction of rotation, the polarity of *only* the armature (or field) would have to be reversed, as shown in Figure 7.5(b).

## Shunt-Wound Motors

In the **shunt-wound motor**, the armature and field windings are connected in parallel [Figure 7.5(d)]. With this configuration, the current in the field is dependent only on the supply voltage. In other words, the field flux is not affected by variations in current due to the CEMF. This results in a motor with a more natural speed regulation.

The torque-speed curve of Figure 7.5(f) graphically shows the shunt-wound motor's characteristics. Notice that both the stall torque and no-load speed values are relatively low when compared to the series motor. Also, the top portion of the curve tends to be more horizontal. This (more) horizontal region is the area of good speed regulation. An increase in torque [ $\Delta T$  in Figure 7.5(f)] reduces the motor's speed by only  $\Delta S$ . In practice, this means a shunt-wound motor will slow down somewhat when the load increases and speed up somewhat when the load is reduced, but on average, a ¼-hp shunt motor will change speed only 15% from a no-load to a rated full-load operation.

Because the shunt motor tends to run at a relatively constant speed, it has traditionally been used in such applications as fans, blowers, conveyer belts, and machine tools. The shunt motor can be designed to operate at almost any torque-speed combination at a given line voltage by specifying the number of turns in the field windings. For a fixed line voltage, the speed can be controlled (within limits) by inserting a rheostat in series with the field winding. As the resistance is increased, the current to the field is reduced, which reduces the field flux and [according to the motor-speed equation (7.6)] speeds up the motor. Speed control can also be achieved by adjusting the line voltage directly. As is the case with the series motor, the shunt motor is reversed by reversing the polarity of only the armature or field coil, but not both (usually it's the armature). The circuit for a reversed shunt motor is shown in Figure 7.5(e).

**EXAMPLE 7.3**

An old 90-Vdc shunt motor on a conveyer belt needs to be replaced. The identification plate on the old motor is unreadable, but you know it was turning at about 1750 rpm. Using your ingenuity and common measuring instruments, determine the specifications for a new motor and select one from the list of Figure 7.4.

**SOLUTION**

The problem comes down to specifying the horsepower, and to know that you need to know how much torque the old motor was supplying. Using a torque wrench, or any torque-measuring device, you turn the conveyer belt shaft and determine it requires about 2 ft · lb to keep it moving.

To convert this data into horsepower requires two formulas;

$$P = TS \quad (7.8)$$

where

$P$  = power

$T$  = torque

$S$  = motor speed

$$1 \text{ hp} = 33,000 \frac{\text{ft} \cdot \text{lb}}{\text{min}} \quad (7.9)$$

where

hp = horsepower\*

First, use Equation 7.8 to calculate the power required to run the conveyer belt (*revolutions* must be converted to *radians*):

$$P = TS = 2 \text{ ft} \cdot \text{lb} \times \frac{1750 \text{ rev}}{\text{min}} \times \frac{2\pi \text{ rad}}{\text{rev}} = 21,990 \frac{\text{ft} \cdot \text{lb}}{\text{min}}$$

Now use Equation 7.9 to convert this result into horsepower:

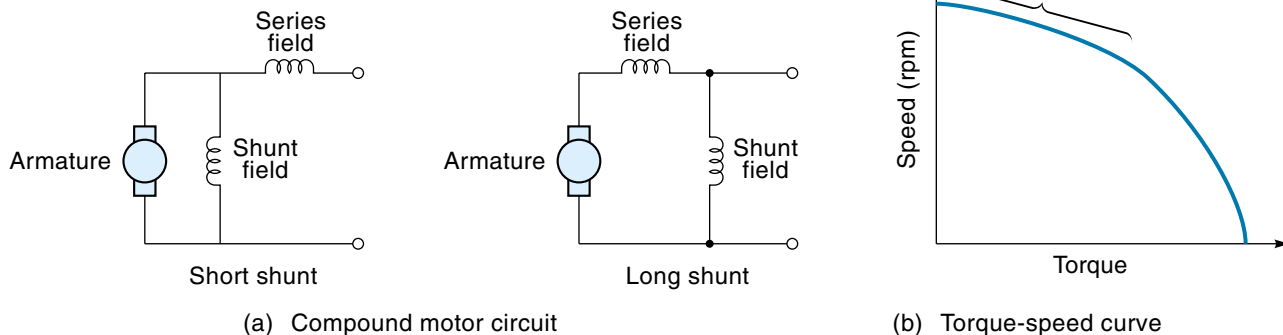
$$\frac{21,990 \text{ ft} \cdot \text{lb}}{\text{min}} \times \frac{1 \text{ hp} \cdot \text{min}}{33,000 \text{ ft} \cdot \text{lb}} = 0.666 \text{ hp}$$

So, based on our calculations, we need a motor that rotates at 1750 rpm with at least 0.666 hp. Looking down the list of Figure 7.4, we see a ¾-hp model with a rated speed of 1750, which should work fine.

\*The unit of horsepower (hp) was developed by James Watt, who determined that a horse could do 33,000 ft-lb of work per minute lifting coal from a coal pit.

**Figure 7.6**

A compound DC motor.



### Compound Motors

The **compound motor** has both shunt and series field windings, although they are not necessarily the same size. There are two configurations of the compound motor, the short shunt and the long shunt, as shown in Figure 7.6(a). Typically, the series and shunt coils are wound in the same direction so that the field fluxes add. The main purpose of the series winding is to give the motor a higher starting torque. Once the motor is running, the CEMF reduces the strength of the series field, leaving the shunt winding to be the primary source of field flux and thus providing some speed regulation. Also, the combination of both fields acting together tends to straighten out (linearize) a portion of the torque-speed curve [Figure 7.6(b)].

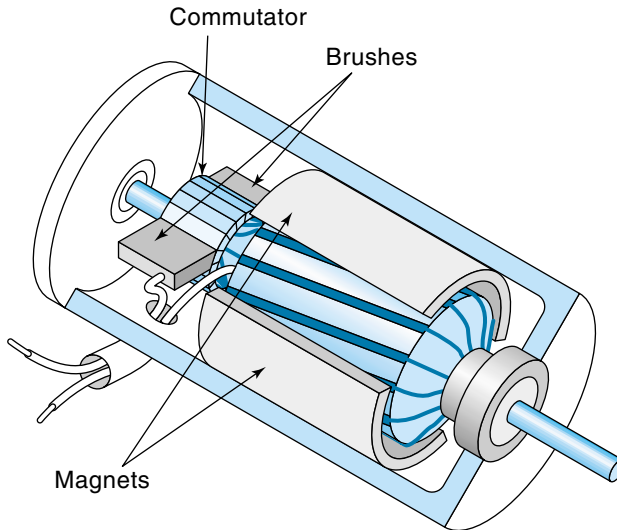
The motor discussed so far, where the fields add, is called a **cumulative compound motor**. Less common is the **differential compound motor**, where the field coils are wound in opposite directions. The differential compound motor has very low starting torque but excellent speed regulation. However, because it can be unstable at higher loads, it is rarely used. The compound motor direction of rotation is reversed by reversing the polarity of the armature windings.

## 7.3 PERMANENT-MAGNET MOTORS

**Permanent-magnet (PM) motors** use permanent magnets to provide the magnetic flux for the field (see Figure 7.7). The armature is similar to those in the wound-field motors discussed earlier. Three types of magnets are used: (1) The Alnico magnet (iron based alloy) has a high-flux density but loses its magnetization under some conditions such as a strong armature field during stalled operation; (2) ferrite (ceramic) magnets have a low-flux density, so they have to be larger, but they are not easily demagnetized; (3) the newer, so-called rare-earth magnets, made from samarium-cobalt or neodymium-cobalt, have the combined desirable properties of high-flux density and high resistance to demagnet-

**Figure 7.7**

Cut-away diagram of a permanent-magnet motor. (In a real motor, the brushes would be pushed against the commutator with a spring mechanism.)



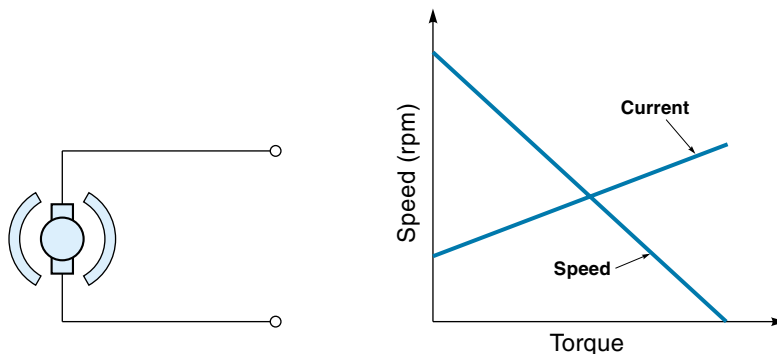
ization. At the present time, the size of PM motors is limited to a few horsepower or less. Small PM motors are used extensively in office machines such as printers and disk drives, toys, equipment such as VCRs and cameras (for zoom and autofocus), and many places in industry. Larger PM motors are used in control systems such as industrial robots.

### Relationship Between Torque and Speed

The fact that the field flux of a PM motor remains constant regardless of speed makes for a very linear torque-speed curve. This is very desirable for control applications because it simplifies the control equations. Figure 7.8 shows a typical PM motor symbol and torque-speed curve. Notice that the “curves” are straight lines for both speed and current. The absence of field coils is apparent in the schematic [Figure 7.8(a)],

**Figure 7.8**

A permanent-magnet motor.



(a) PM motor symbol

(b) PM motor torque-speed curve

which shows the applied voltage feeding only the armature. The PM motor is easily reversed by reversing the polarity of the applied voltage. Example 7.4 illustrates how the torque-speed curve can be used to predict the performance of a motor under any load condition.

#### EXAMPLE 7.4

Figure 7.9(a) shows the torque-speed curve of a PM motor. Find the speed and motor current for the following:

- a. No-load and stall conditions
- b. Lifting a 10-oz load with a 2-in. radius pulley
- c. A motor driving a robot arm with a weight

#### SOLUTION

- a. If voltage is applied to the motor with no load attached to the shaft, the motor would turn at its no-load speed of 1000 rpm. On the other hand, if the shaft was clamped so it could not turn, the motor would exert the stall torque of 100 in. · oz on the clamp and draw 260 mA of current.
- b. A 10-oz weight is hung from a 2-in. radius pulley. The pulley is on the motor shaft [Figure 7.9(b)]. Torque equals force times distance, so the motor torque required to lift the weight is

$$2 \text{ in.} \times 10 \text{ oz} = 20 \text{ in.} \cdot \text{oz}$$

From the graph, we can see that, at a torque of 20 in. · oz, the speed has declined to 800 rpm, and the current is up to 125 mA.

- c. The motor is attached to a 12-in. robot arm (weighing 10 oz). On the end of the arm rests an 8-oz apple [Figure 7.9(c)]. The total load on the motor is calculated as shown below. Note the arm is considered to be a point mass of 10 oz at its center of gravity, which is 6 in. from the motor shaft.

$$\underbrace{(6 \text{ in.} \times 10 \text{ oz})}_{\text{Arm}} + \underbrace{(12 \text{ in.} \times 8 \text{ oz})}_{\text{Apple}} = 156 \text{ in.} \cdot \text{oz}$$

From the graph, we see that 156 in. · oz exceeds the stall torque of 100 in. · oz, so the motor will not be able to lift this load at all. One solution to the problem would be to insert a gear train of, say, 5 : 1 between the motor and load, [Figure 7.9(d)]. Now the torque required of the motor is only one-fifth of what it was, or:

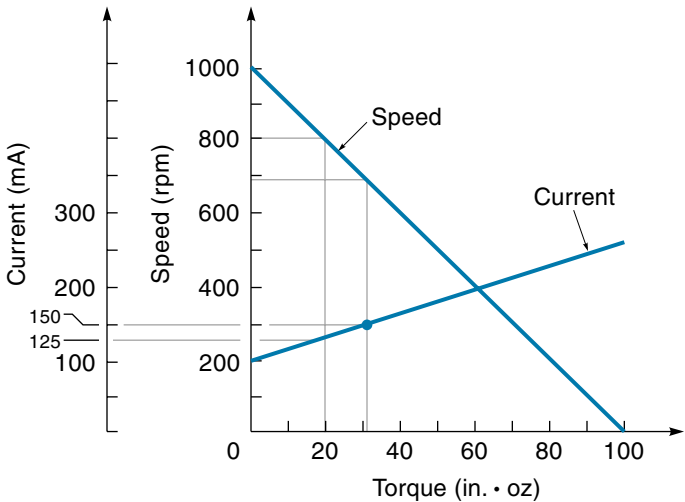
$$\frac{156 \text{ in.} \cdot \text{oz}}{5} = 31.2 \text{ in.} \cdot \text{oz}$$

Which is within the torque range of the motor, so it will rotate at 690 rpm and require a current of 150 mA. However, because of the gear train, the load will rotate at only one-fifth of the motor shaft, or

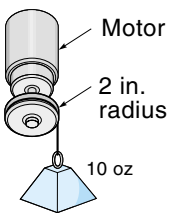
$$\frac{690 \text{ rpm}}{5} = 138 \text{ rpm}$$

Real-world loads are not always constant. In this case, the action of gravity causes the load to be greatest when the arm is horizontal and zero when the arm is vertical, so we would expect the motor speed to increase and decrease with each revolution.

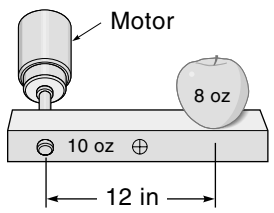
**Figure 7.9**  
Torque–speed curve  
and hardware setups  
for Example 7.4.



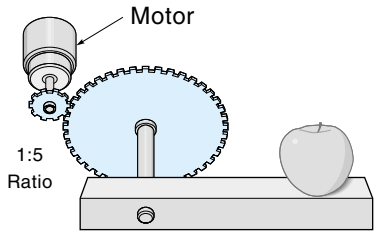
(a)



(b)



(c)



(d)

A single torque-speed curve represents the motor performance for a particular voltage. Control motors are not usually operated at a single voltage because varying the voltage is a way of controlling the power (and therefore speed). Figure 7.10 shows a family of torque-speed curves, where each curve represents the motor response for a different voltage. As the voltage increases, the stall torque and no-load speed also rise, but saying that speed increases with voltage is too simple. What is really increasing is the torque, and in most cases, increased torque results in increased speed.

### EXAMPLE 7.5

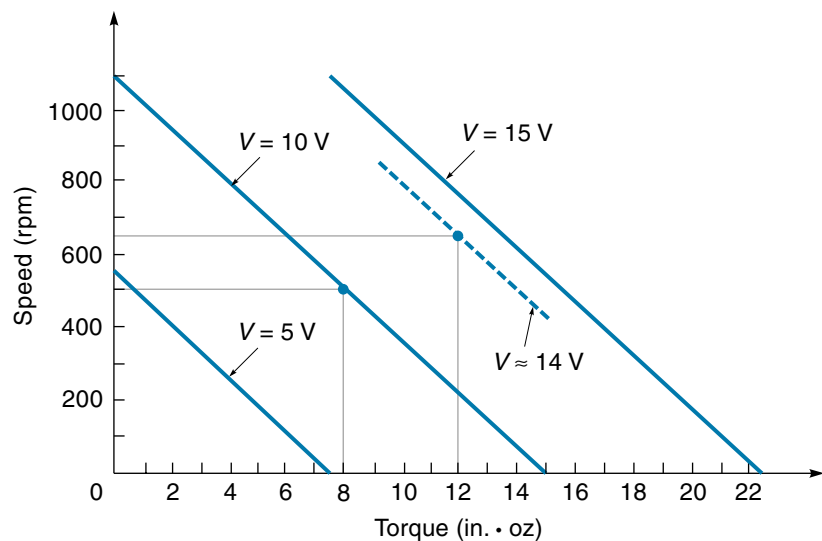
A small PM motor is used on a tape deck. During rewind the motor runs at 500 rpm at 10 V. To shorten the rewind time, we want the motor to run at 650 rpm. It has been estimated that the increased speed will increase the load torque by 50%. Based on that estimate, determine what voltage must be applied to the motor to make it run at 650 rpm (use the torque-speed curve of Figure 7.10).

### SOLUTION

On the torque-speed curve of Figure 7.10, identify the existing operating point of the motor on the 10-V curve at 500 rpm. The motor is supplying 8 in. · oz at that speed and voltage; for the faster rewind, the motor needs to run at 650 rpm supplying 12 in. · oz (12 in. · oz is 50% greater than 8 in. · oz). The new operating point does not lie directly on one of the voltage curves, so we must sketch in some parallel lines and estimate the new voltage to be approximately 14 V.

**Figure 7.10**

A family of torque-speed curves (Example 7.5).





**Figure 7.11**  
Small PM motor specifications.  
(Courtesy of Bertsch U.S.)



# BERTSCH U.S. MICROMOTORS

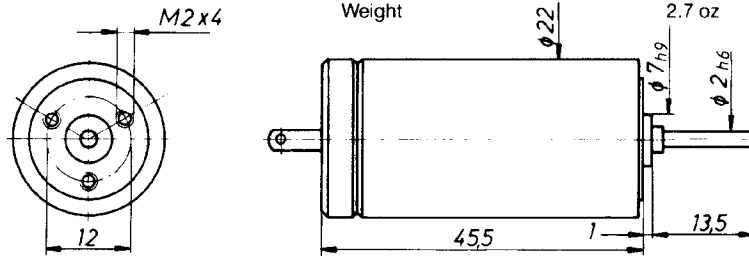
**Series 22-45 · 6 Watts · 10,000 RPM**

Dimensions – mm 1 mm = 0.0394 in

Magnet	AlNiCo	Efficiency	up to 85 %
Commutator	CuAg 0.1, 5 segments	Rotor Moment of Inertia	$20 \times 10^{-3}$ oz-in <sup>2</sup>
Brushes	Graphite	Mechanical Time Const. appr.	20 ms
Bearings	Oilite	Weight	2.7 oz

**Possible Combinations**

Gear Heads	H 30, H 40, P 24
Tacho	20-08
Encoder	HEDS-5500



**Recommended Operational Data**

Speed	max. 10,000 rpm
Torque	0.7 oz-in
Temperature	-4° to +175° F

Type	22-45-30	22-45-20	22-45-16	22-45-14	22-45-12	22-45-10	22-45-08	
Voltage	3	6	9	12	18	24	36	V
No load speed	7,500	10,000	10,100	10,000	9,700	9,900	9,700	rpm
Resistance	0.8	2.5	4.5	6	14	28	65	Ohms
Stall torque	1.89	1.80	2.19	3.07	2.95	2.63	2.12	oz-in
No load current appr.	65	55	45	35	25	20	18	mA
Recommended current	1000	800	600	450	300	250	150	mA
Counter-EMF	0.51	0.77	1.15	1.56	2.34	3.15	4.78	mV/rpm
Speed/Torque Gradient	3980	5560	4510	3270	3420	3780	4740	rpm/oz-in

**Gear Heads H 30 and H 40 with Oilite Bearings**

**Standard Gear Ratios**

<b>H 30:</b>	20 : 1	60 : 1	200 : 1	600 : 1	2500 : 1
<b>H 40:</b>	40 : 1	100 : 1	200 : 1	500 : 1	1000 : 1    2000 : 1

Gear Head	H 30	H 40	
Max. torque	30	42	oz-in
Max. Motor Speed	5000	7500	rpm
Max. Power	1	2	W
Max. Axial Load	36	54	oz
Max. Radial Load	36	36	oz
Weight with motor	4.2	5	oz

Figure 7.11 shows an example of an “instrument-type” PM motor. This small motor (about 1 in. in diameter) is available in seven models from 3 to 36 V. Consider the 3-V model (22-45-30), which has a no-load speed of 7500 rpm and a stall torque of 1.89 in. · oz [Figure 7.11]. The torque-speed curve could be constructed from these figures (see Example 7.6). This motor is available with small **gearheads**, which are gear

trains that are designed to connect directly to the motor. In this case, gear ratios from 20 : 1 to 2000 : 1 are available.

### EXAMPLE 7.6

For motor 22-45-30 [Figure 7.11],

- a. Draw the torque-speed curve.
- b. Determine the speed for a load torque of 1 in. · oz.

### SOLUTION

- a. From Figure 7.11, we see that motor 22-45-30 has a no-load speed of 7500 rpm and a stall torque of 1.89 in. · oz. To draw the torque-speed curve, draw the axes and establish an appropriate scale so that 7500 is on the  $y$ -axis and 1.89 is on the  $x$ -axis. Connect these two points with a straight line as shown in Figure 7.12 (remember, only a PM motor has a straight line).
- b. Using the torque-speed curve, we can now determine the speed when the motor is delivering 1 in. · oz of torque. From Figure 7.12, we see that the speed would be 3400 rpm.

## Circuit Model of the PM Motor (Optional)

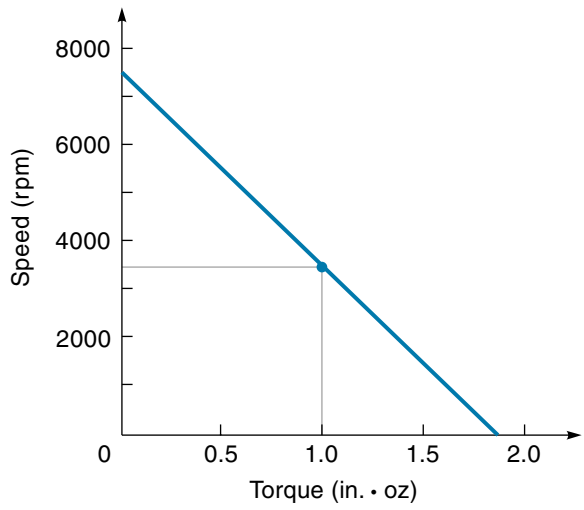
[*Note:* This section can be omitted without affecting the comprehension of the rest of this chapter.]

So far we have been examining the performance of the PM motor and found that it is dependent on both the electrical and mechanical conditions present. To understand *all* factors that affect the motor's performance in a particular situation, examining a comprehensive motor-load mathematical model is helpful. This model is simply an equation, where each term represents some electrical or mechanical condition. In this section, we will look at the basic equations that describe motor-system performance.

Figure 7.13 shows a simple model of the PM motor driving a mechanical load. The equivalent circuit of the motor is a series circuit consisting of an inductance ( $L_A$ ) from the armature windings, a resistance ( $R_A$ ) to account for wire and commutator resistance, and a voltage source CEMF. Recall that the CEMF is an internally generated voltage that increases with motor speed. The polarity of the CEMF is such that it subtracts from the line voltage ( $V_{in}$ ), creating an effective armature voltage. The inductance ( $L_A$ ) limits how fast the armature current can change but doesn't affect the steady-state operation. Writing Kirchhoff's equation around the loop of Figure 7.13 (for steady state), we get

**Figure 7.12**

Torque–speed curve for motor 22-45-30.



$$V_{in} = (R_A \times I_A) + \text{CEMF}$$

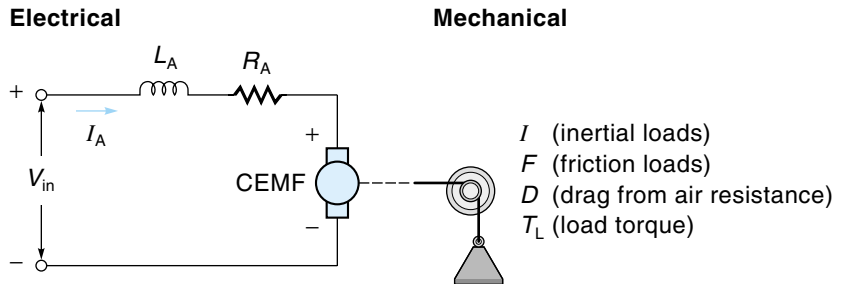
Solving for  $I_A$ , we get

$$I_A = \frac{V_{in} - \text{CEMF}}{R_A} \tag{7.10}$$

Therefore, for steady-state operation, the motor current is directly proportional to the difference between the applied voltage and the CEMF. *The value of motor current is particularly important because it is the electrical parameter that most directly relates to mechanical torque.* For the PM motor, the relationship between torque and current is very simple, as expressed in equation 7.11 (given on next page).

**Figure 7.13**

Equivalent circuit of a PM motor with a mechanical load.



$$T = K' I_A \quad (7.11)$$

where

$T$  = mechanical torque

$K'$  = motor constant

$I_A$  = armature current

The motor, together with its load, is a mechanical as well as an electrical system. Figure 7.13 indicates some of the mechanical factors that will effect the motor's performance. These include the **moment of inertia** ( $I$ ), the bearing friction ( $F$ ), and the drag ( $D$ ). The drag, called *windage*, comes from air resistance on moving parts. All these factors cause resisting torques that the motor must overcome. The moment-of-inertia factor opposes any change in speed, the bearing friction is a constant, and the air drag increases with speed. Equation 7.12 describes the mechanical torque on a *free-running motor*, that is, a motor not connected to any external load:

$$T = (I_m \times \Delta S) + F_m + (D \times S) \quad (7.12)$$

where

$T$  = torque needed to spin a motor with no load

$I_m$  = moment of inertia of the armature and shaft

$S$  = motor speed

$\Delta S$  = change in the motor speed per unit time (acceleration)

$F_m$  = friction in the motor from bearings

$D$  = drag factor from wind resistance

The effect of the moment of inertia is only present when the motor speed is increasing or decreasing. The larger the  $I$  factor, the more the speed resists change. Consequently, the system response time is slowed by the moment of inertia, whereas the steady-state speed is unaffected.

When the motor is connected to an external load, an additional set of  $I_L$ ,  $F_L$ , and  $T_L$  influences the motor speed. The external load will certainly have its own moment of inertia and friction, and there is now an external **load torque** ( $T_L$ ), which is the only useful work being done by the motor and the reason for it being there in the first place. The complete mathematical picture of what the motor has to drive is shown in Equation 7.13:

$$T = \underbrace{(I_M \times \Delta S) + F_M + (D \times S)}_{\text{Motor}} + \underbrace{(I_L \times \Delta S) + F_L + (D_L \times S) + T_L}_{\text{Load}} \quad (7.13)$$

where

$I_L$  = load moment of inertia

$F_L$  = load friction

$D_L$  = wind-resistance drag of the load

$T_L$  = load torque (useful work)

If there is a gear train between the motor and load, the moment of inertia and friction will be affected. In most cases, the motor is geared down to the load, so from the motor's point of view, the load values of  $I_L$ ,  $F_L$ ,  $D_L$ , and  $T_L$  are reduced. In summary, the motor's actual performance is determined by its internal characteristics *and* the nature of the load.

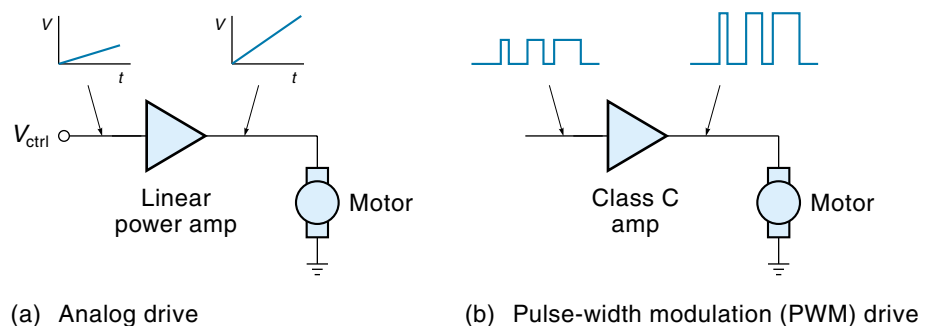
## 7.4 DC MOTOR-CONTROL CIRCUITS

In this section, we discuss the two basic approaches for controlling the speed of a DC motor. The term *speed control* is somewhat inaccurate because, technically speaking, the motor converts electrical energy into torque, not speed (the precise speed is determined by *both* the motor torque and the mechanical load). Still, in general, it is safe to say that increased voltage will result in increased speed.

To drive the motor, an interface circuit is required to convert the low-level motor-control signal from the controller into a signal strong enough to run the motor. The classical way to do this is with an **analog drive**. In this method, a linear power amplifier amplifies the drive signal from the controller and gives the motor a “strengthened” analog voltage [Figure 7.14(a)]. A DAC (digital-to-analog converter) would be required if the controller is digital.

The other technique for controlling a DC motor is **pulse-width modulation (PWM)**. In this system, power is supplied to the motor in the form of DC pulses of a fixed voltage [Figure 7.14(b)]. The width of the pulses is varied to control the motor speed. The wider the pulses, the higher the average DC voltage, so more energy is available to the motor. The frequency of the pulses is high enough that the motor's inductance averages them, and it runs smoothly. This system has two advantages over the analog drive: (1) The power amplifier can be of the efficient class C type, and (2) the DAC is not needed because the amplifier is either on or off and can be driven directly with a digital signal.

**Figure 7.14**  
Methods of  
speed control  
for DC motors.



## DC Motor Control Using an Analog Drive

The analog-drive system of DC motor control uses a linear-power amplifier to drive the motor. The amplifier is the interface between the controller and the motor. Typically, it is a current amplifier, meaning its primary function is to boost the current; the output voltage may or may not be larger than the input voltage.

The simplest analog-drive circuit is a class A amplifier using a single power transistor. The circuit could be either the *common emitter* (CE) configuration, which gives current and voltage gain, or the *common collector* (CC) configuration, which gives only current gain; these circuits are shown in Figure 7.15. The operation of both circuits is similar; when the base voltage ( $V_B$ ) is increased (beyond the forward-bias voltage), the transistor begins to turn on and let the collector current ( $I_C$ ) flow. The collector current is 30–100 times greater than the base current, depending on the gain ( $h_{fe}$ ) of the transistor. Once the transistor starts to conduct,  $I_C$  increases with  $V_B$  more or less linearly. Note that all of  $I_C$  goes through the motor, providing the drive current.

The problem with this arrangement is that class A amplifiers are very inefficient. Consider the case of a 12-V, 2-A motor connected as shown in Figure 7.16(a). The DC power available is 12 V. When the transistor is all the way on, it is acting like a closed switch, so the entire 12 V is applied across the motor. When the transistor is half-on, it is acting like a resistor in series with the motor, reducing  $I_C$  by one-half, in this case to 1 A. Now, only 6 V is dropped across the motor, and the remaining 6 V is dropped across the transistor. The power being drawn from the supply is

$$\begin{aligned} P &= IV \\ &= 1 \text{ A} \times 12 \text{ V} = 12 \text{ watts (W)} \end{aligned}$$

whereas the power being dissipated by the transistor is

$$\begin{aligned} P &= VI \\ &= 6 \text{ V} \times 1 \text{ A} = 6 \text{ W} \end{aligned}$$

**Figure 7.15**

Analog-drive output configurations for a DC motor (conventional current flow).

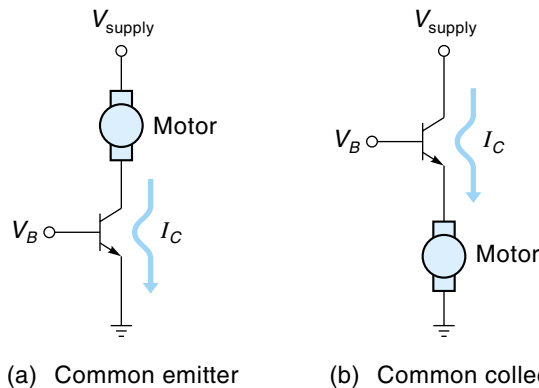
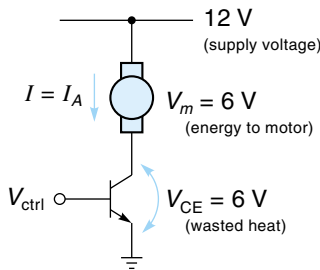
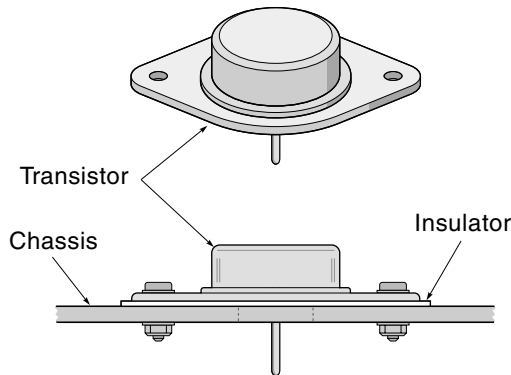


Figure 7.16

Power transistors.



(a) Circuit conditions when a transistor is half-on



(b) Power transistor mounting

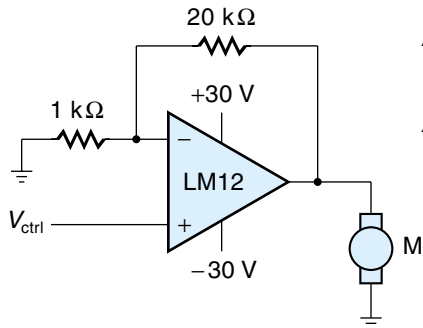
Half the power (6 W) of this system is wasted in the form of heat by the transistor when the motor is running at half-speed. In many control systems, the motor may be running with an average speed of even less than half. For smaller motors, these losses may be acceptable, but for larger motors or battery operation a more efficient arrangement is needed.

The circuit just described uses a power transistor as the motor driver amp. As shown in Figure 7.16(b), power transistors and other power-driver circuits are made so that they can be mounted securely on the chassis. The purpose of the mounting is to provide a path for the heat to escape. A power transistor that is not mounted (or mounted incorrectly) cannot handle anywhere near its rated power. Typical power transistors can carry up to 60 A and dissipate as much as 300 W, and special models can go much higher. (Power transistors and heat considerations are covered in some detail in Chapter 4.) Other motor drivers besides the power transistor include the power IC, the Darlington power transistor, and the power MOSFET; these are described next.

The power IC driver is a single-package DC amplifier with a relatively high current output. An example is the LM12 (National Semiconductor) shown in Figure 7.17. This

Figure 7.17

LM12 power operational amplifier (National Semiconductor).

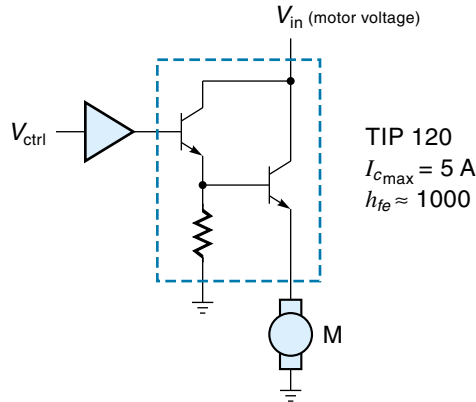


$$A_v = \frac{R_f}{R_i} + 1$$

$$A_v = \frac{20 \text{ k}\Omega}{1 \text{ k}\Omega} + 1 = 21$$

**Figure 7.18**

A DC motor drive using a power Darlington transistor.



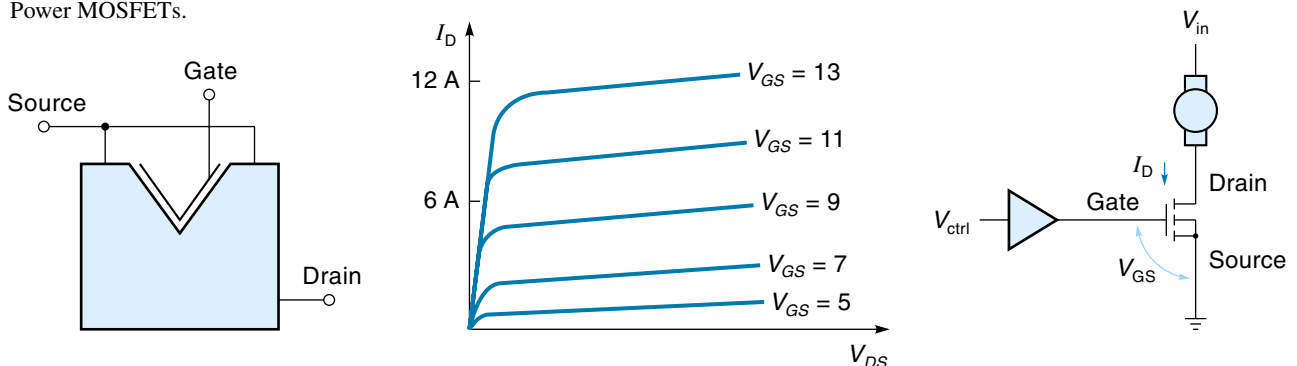
high-power operational amplifier can supply up to 13 A with a maximum voltage of  $\pm 30$  V. As in any op-amp circuit, feedback resistors are added to adjust the gain to any desired value. The voltage gain for the circuit of Figure 7.17 is 21 [ $A_v = (R_f/R_i) + 1$ ].

Figure 7.18 shows a motor-driver circuit using a power Darlington transistor. The Darlington configuration consists of two CC amplifiers connected in such a way that the first transistor directly drives the second. Although the voltage gain is only 1 (maximum), the current gain can be very high. The transistor shown in Figure 7.18 is a TIP 120, which has a current gain ( $h_{fe}$ ) of 1000 and a maximum output current of 5 A. The motor must be placed in the emitter path of the output transistor. A separate small-signal amplifier, probably an op-amp, would be needed to provide any voltage gain required.

Another device capable of providing analog drive for a motor is the power MOSFET, known by such names as VFET, TMOS, and HEXFET. The shape of its construction allows for large currents [Figure 7.19(a)]. These FETs are usually designed to operate in the enhancement mode, which means the biasing voltage is always posi-

**Figure 7.19**

Power MOSFETs.



(a) Power MOSFET construction

(b) Output curves

(c) Motor-driver circuit



**Figure 7.20**

Reversing motor rotation with negative voltage (conventional current).

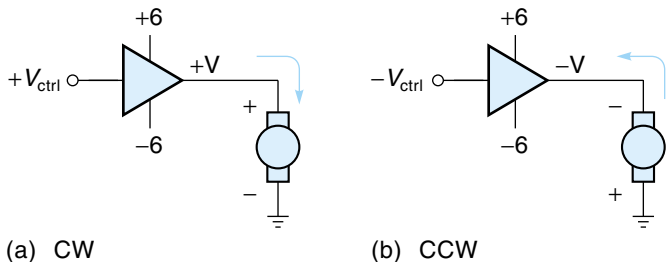


Figure 7.19(b) shows a typical set of curves for the power MOSFET. Notice the output current ( $I_D$ ) is 0 A when the input voltage ( $V_{GS}$ ) is in the 0-5-V range but then climbs to 12 A when  $V_{GS}$  rises to 13 V. Figure 7.19(c) shows the basic motor-driver circuit using a power MOSFET. In this case, the motor is in series with the drain, which means the FET will provide both voltage and current gain. The gate voltage is supplied from an op-amp circuit that is designed to interface the controller with the FET.

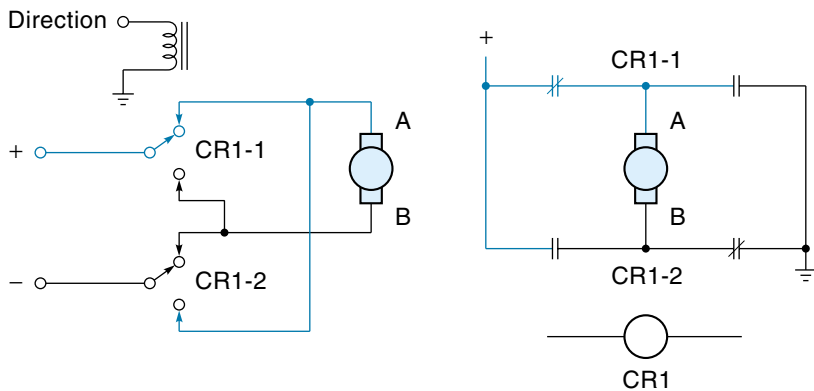
**Reversing the PM Motor**

To reverse the rotation direction of the PM motor, the polarity of the applied voltage must be reversed. One way to accomplish this is to have a motor-driver amp capable of outputting a positive and negative voltage (Figure 7.20). When the drive voltage is positive with respect to ground, the motor turns clockwise (CW). When the drive voltage is negative with respect to ground, the voltage polarity at the motor terminals reverses, and the motor rotates counterclockwise (CCW). The LM12 power op-amp (Figure 7.17) is capable of providing positive and negative output voltages.

In many applications, the drive amplifier cannot output both positive and negative voltages, in which case a switching circuit must be added to reverse the motor. One approach is to use a double-pole relay (Figure 7.21). When the relay contacts are up, the positive voltage is connected to terminal A of the motor, and terminal B is connected

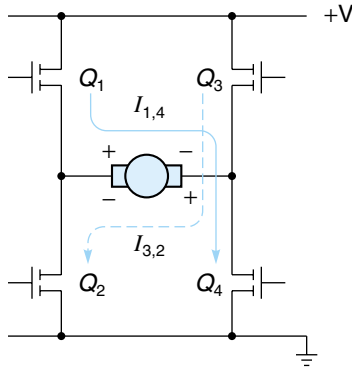
**Figure 7.21**

Reversing a motor with relay switching (two schematic representations).



**Figure 7.22**

Reversing a DC motor with solid-state switching (conventional current).

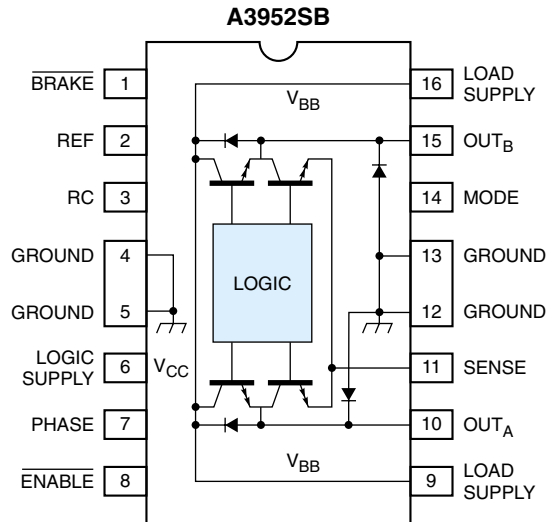


to the negative voltage. When the relay contacts are down, the positive voltage is connected to terminal B, and terminal A goes to the negative voltage, thus effectively reversing the polarity.

Forward-reverse switching can also be done with solid-state devices. Figure 7.22 shows a motor-reversing circuit that uses four FETs. When  $Q_1$  and  $Q_4$  are on, the current  $I_{1,4}$  causes the motor to turn clockwise. When  $Q_2$  and  $Q_3$  are on, the current  $I_{3,2}$  flows in the opposite direction and causes the motor to turn counterclockwise. The entire switching operation can be performed by a single IC, such as the Allegro A3952 shown in Figure 7.23. This IC contains four separate driver transistors that are controlled by internal logic to operate in pairs (in the manner of Figure 7.22). The A3952 controls a motor-supply voltage of up to 50 V with up to 2 A of output current.

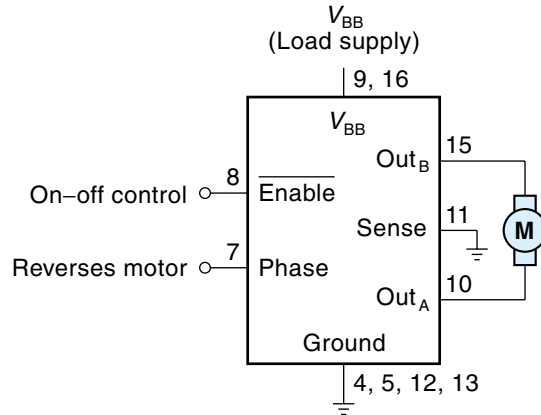
**Figure 7.23**

A full-bridge PWM motor driver. (Courtesy of Allegro Micro Systems Inc.)



**Figure 7.24**

Application of the Allegro A3952.



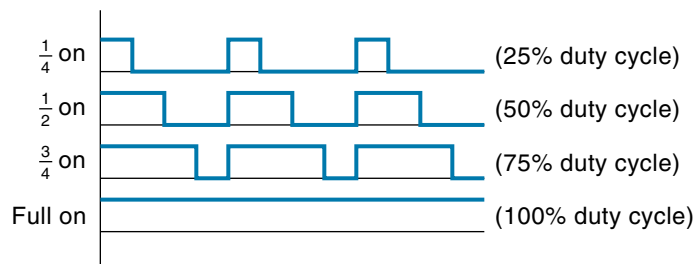
The A3952 is very simple to use, as illustrated in Figure 7.24. Basically, the IC connects the supply voltage ( $V_{BB}$ ) and ground to the motor through pins  $Out_A$  and  $Out_B$ . The polarity of the motor voltage (that is, motor direction) is controlled by the Phase input, and the Enable input can be used to turn the motor on and off. Both Phase and Enable inputs are digital TTL-compatible signals.

### DC Motor Control Using Pulse-Width Modulation

Pulse-width modulation is an entirely different approach to controlling the torque and speed of a DC motor. Power is supplied to the motor in a square wavelike signal of constant magnitude but varying pulse width or duty cycle. **Duty cycle** refers to the percentage of time the pulse is high (per cycle). Figure 7.25 shows the waveforms for four different speeds. For the slowest speed, the power is supplied for only one-quarter of the cycle time (duty cycle of 25%). The frequency of the pulses is set high enough to ensure that the mechanical inertia of the armature will smooth out the power bursts, and the motor simply turns at a constant velocity of about one-quarter speed. For a 50% duty cycle (power on one-half the time), the motor would turn at about half speed, and so on. In real life, nonlinear factors cause the motor to go slower than the straight

**Figure 7.25**

PWM waveforms.



proportions suggest, but the principle still holds—that is, the speed of a motor can be regulated by pulsing the power.

PWM offers two distinct advantages over analog drive. First, it is digital in nature—the power is either on or off, so it can be controlled directly from a computer with a single bit, eliminating the need for a DAC. Second, the drive amplifier can use efficient class C operation. The class C amplifier is efficient because the conditions that cause power dissipation are minimized. Consider the circuit shown in Figure 7.26. A control voltage ( $V_{\text{ctrl}}$ ) of 2 V will turn on  $Q_1$  all the way (saturation), and  $I_C$  flows freely through the motor. That is,  $Q_1$  acts like a closed switch, making  $V_{\text{CE}}$  near 0 V and so dropping the entire line voltage ( $V_{\text{in}}$ ) across the motor. We can calculate the power dissipated by the transistor using the power equation:  $P = IV$ . Applying this to  $Q_1$ ,

$$P = I_C \times V_{\text{CE}}$$

If  $V_{\text{CE}}$  is near 0 V, then

$$P = I_C \times 0 \text{ V} = 0 \text{ W}$$

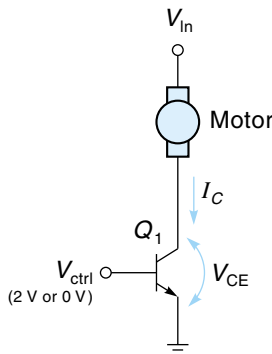
This tells us that almost no power is lost by the transistor when  $I_C$  is flowing. On the other hand, when  $V_{\text{ctrl}}$  is 0 V,  $Q_1$  is turned completely off, thus reducing  $I_C$  to near 0 A. Recalculating the power under these conditions,

$$P = 0 \text{ A} \times V_{\text{CE}} = 0 \text{ W}$$

Again, theoretically no power is dissipated by the transistor. Therefore, if the transistor is operated so that it is either all the way on or all the way off, we have a very efficient amplifier. This is class C operation, and it can be used for PWM. In practice, switching transistors do have losses because of leakage and because they can't turn on (or off) instantaneously. Therefore, a little dissipation occurs with each transition, which means that the higher the frequency, the hotter the transistor gets!

**Figure 7.26**

Class C amp operation using a power transistor (conventional current).



The driver amplifier circuit used for PWM is essentially the same as for an analog drive. The power transistor, power Darlington, power MOSFET, and power IC amp can all be used. One important difference from analog drive is that PWM amplifiers do not have to be linear; this tends to make the PWM amp less complicated. Biasing is designed so that the amplifier is normally off. The input signal will look like that shown in Figure 7.25. The magnitude of the input must be high enough to ensure that the amp will turn on all the way during the pulses. If not, the resulting inefficiency may cause the amplifier to fail from overheating.

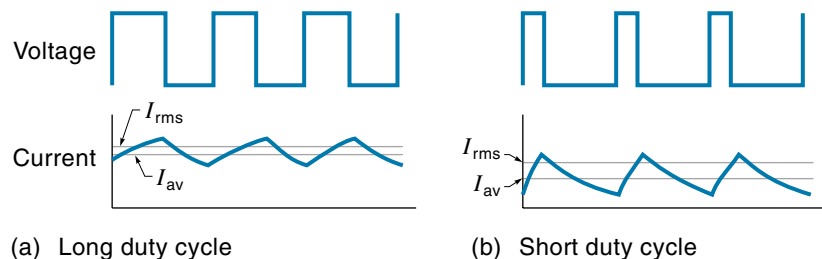
The Allegro A3952 (Figures 7.23 and 7.24) is ideally suited to provide a PWM drive. The load supply terminal ( $V_{BB}$ ) is connected to a DC source and provides the motor voltage. A TTL-level PWM waveform (such as in Figure 7.25) is connected directly to the Enable input. Recall that the Enable input turns on and off the motor voltage.

Some problems are associated with pulsing power to a DC motor. In fact, we usually try to avoid rapidly turning on and off an inductive load like a motor because of the large voltage spikes generated when the interrupted armature current has no place to go. Recognizing this problem, the PWM circuit includes *free-wheeling* or *flyback diodes* to provide a nondestructive return path for the current. Figure 7.23 shows four flyback diodes—one across each transistor. Each time the voltage is withdrawn from the motor, a brief voltage spike will appear at the motor terminals. The diodes redirect this energy back into the power source.

Because the PWM drive current is not a steady DC value, a unique kind of inefficiency arises, which the analog drive does not have. Figure 7.27 shows the PWM input voltage and current for two different duty cycles. Notice that the current waveform looks more like a sawtooth than a square wave. The current is trying to follow the voltage waveform, but the inductance of the motor restricts the current from changing very fast. As the pulses get shorter, the relative peak-to-peak value of the current increases. Like any periodic waveform, this current waveform has both an average value ( $I_{av}$ ) and an rms value ( $I_{rms}$ ), but note that the difference between the two values gets larger as the pulses get shorter. Here is where the inefficiency occurs because the armature heat losses are proportional to the (higher)  $I_{rms}$  value and the mechanical torque of the motor is proportional to the (lower)  $I_{av}$  value. These two relationships are mathematically described in equations 7.14 and 7.15.

**Figure 7.27**

Difference of  $I_{ave}$  and  $I_{rms}$  in PWM waves.



$$T = I_{\text{av}} \times K_t \quad (7.14)$$

where

$T$  = motor torque

$I_{\text{av}}$  = average armature current

$K_t$  = motor constant

$$P_d = I_{\text{rms}}^2 \times R_a \quad (7.15)$$

where

$P_d$  = power dissipated through the armature

$I_{\text{rms}}$  = value of the rms component of the current

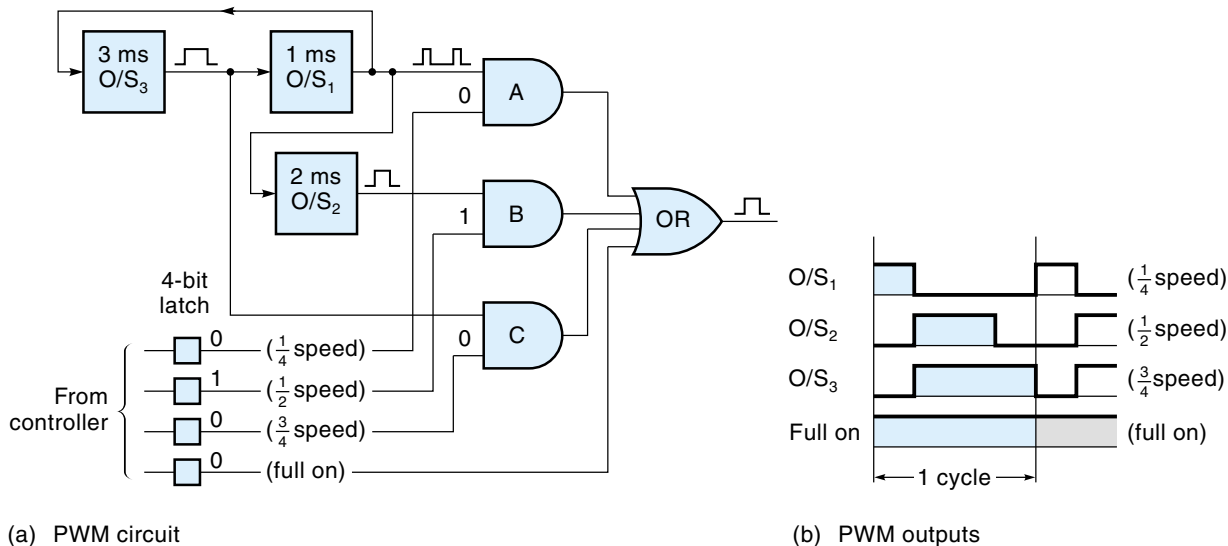
$R_a$  = armature resistance

In summary, as the PWM pulses get shorter, the  $I_{\text{rms}}$  component gets proportionately larger, as does the armature losses. The implication here is that slow motor speeds require the duty cycle to be more than a simple proportion; for example, for 25% speed, the duty cycle might have to be 35%.

## PWM Control Circuits

There are many ways to create the PWM waveform, using either hard-wired circuits or software. However, so not to burden the main processor with such a repetitive task, PWM is usually generated with dedicated circuits, or with special programmable timing circuits built-in to the microcontroller. In this section, we examine some different methods that can be used to produce PWM. In all designs, it is the duty cycle, not the frequency, that determines motor speed. The frequency stays constant and is usually in the range of 40 Hz-10 kHz. Lower frequencies can sometimes cause objectional vibrations, and power transfer to the motor declines at higher frequencies due to the inductive reactance of the armature windings.

Figure 7.28 shows a 4-speed PWM control circuit. Operation of the circuit is as follows: Two *one-shots*, O/S<sub>1</sub> and O/S<sub>3</sub>, are connected together to make an oscillator. (A one-shot generates a single pulse when triggered.) The 3-ms one-shot triggers the 1-ms one-shot, which feeds back and triggers the 3-ms one-shot, and so on. The output of this circuit is a series of pulses that are high (5 V) for 1 ms and low for 3 ms. This is the 1/4-speed waveform shown in Figure 7.28(b). Note that the basic period of the waveform has been established at 4 ms. A third one-shot (O/S<sub>2</sub>) creates a 2-ms pulse and is triggered by the “falling edge” of the O/S<sub>1</sub> output. The output of O/S<sub>2</sub> is high for one-half of the 4-ms period and is shown as the 1/2-speed waveform in Figure 7.28(b). By taking the output of the 3-ms O/S<sub>3</sub>, we create a 3/4-speed waveform as shown. Control gates A, B, and C admit one of the three signals into the OR gate. A fourth signal into the OR gate called “full on” (duty cycle = 100%) provides for straight DC to the motor. The control signals could come directly from the computer. The computer would merely send a 4-bit data word to a latching output port, with a 1 in the bit position of the desired speed. The motor would operate at that speed until a new command was sent.



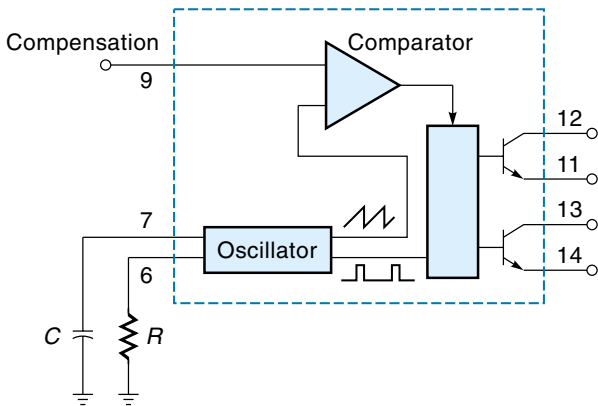
(a) PWM circuit

(b) PWM outputs

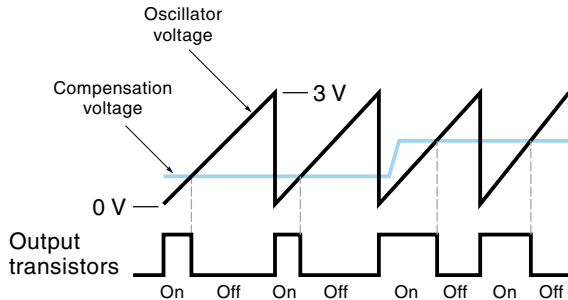
Figure 7.28

Circuit for a 4-speed PWM motor control (shown with 1/2-speed selection).

Some ICs can generate the PWM waveform directly. One such device is the LM3524 (National Semiconductor) shown in the block diagram of Figure 7.29(a). The duty cycle can be varied from 0% to almost 100% by applying a DC voltage called “compensation” on pin 9. An on-board oscillator establishes the PWM frequency in the form of a sawtooth waveform (frequency can be adjusted with an external *R* and *C*). This sawtooth waveform is compared with the compensation voltage in a comparator



(a) Block diagram of a PWM IC (LM3524)



(b) Output transistors turn off when the oscillator voltage reaches compensation

Figure 7.29

Pulse-width modulation IC (LM3524). (Courtesy of National Semiconductor)

amp. At the start of each cycle, the output transistor is turned on; when the sawtooth voltage rises to the value of the compensation voltage, the output transistor turns off and stays off for the rest of that cycle [Figure 7.29(b)]. As the compensation voltage is increased, the output transistor stays on longer, increasing the duty cycle.

One of the most common ways to produce PWM is with programmable hardware built into the microcontroller. For example, PWM can be generated from the MC68HC11 microcontroller by loading two numbers into registers corresponding to the high time and the low time of each cycle. The PWM output is set low while a high-speed counter counts up to the value in the low-time register, then the output is set high while the counter counts up to the value in the high-time register. Interrupt routines automatically take care of the whole PWM process, so the microcontroller is basically free to do other things. Other microcontrollers, such as the MC68HC05B6, come with built-in PWM channels. Each channel can be initialized for a specific period and duty cycle, then the hardware takes over and produces the PWM wave. If the application program needs to change the duty cycle, a new value is simply written to the duty-cycle register.

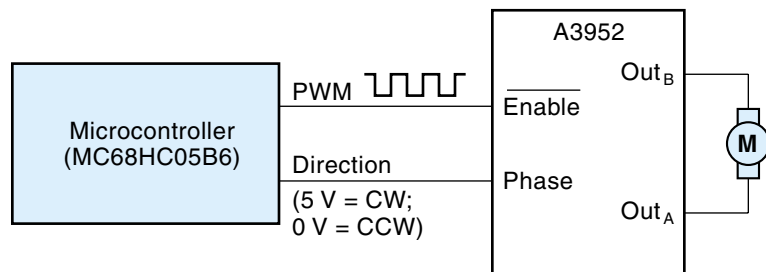
Figure 7.30 shows an MC68HC05B6 microcontroller (Motorola) being used to provide PWM drive to a DC motor. The output of the microcontroller is a 0-5-V logic signal that must be amplified to provide the necessary motor current. In this case, an A3952 (discussed earlier) will be used. Notice that the PWM signal goes to the enable input and the signal that specifies motor direction goes to the phase input.

## DC Motor Control for Larger Motors

DC motors come in all sizes, from small instrument motors requiring a fraction of an ampere to large industrial motors requiring hundreds of amperes. The discussion thus far has assumed that a DC supply voltage was available and that all the control circuit had to do, broadly speaking, was to connect this voltage to the motor. For larger motors—say, 20 A or more—the hardware needed to supply pure DC becomes bulky and expensive. An alternative solution is to drive the DC motor with rectified AC, where no attempt is made to smooth the waveform. A device that is frequently used in this application to provide both rectification and some measure of control is the **silicon-controlled rectifier (SCR)**, which is covered in some detail in Chapter 4. The following paragraph is only a summary of SCR operation.

**Figure 7.30**

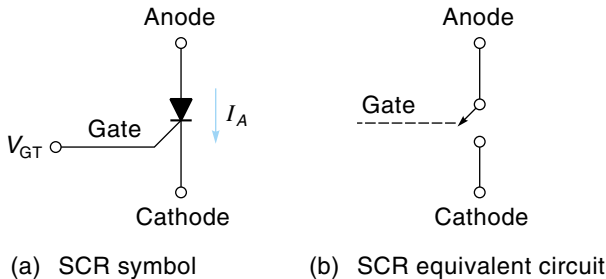
Controlling a DC motor with PWM (from a microcontroller).





**Figure 7.31**

A silicon-controlled rectifier.



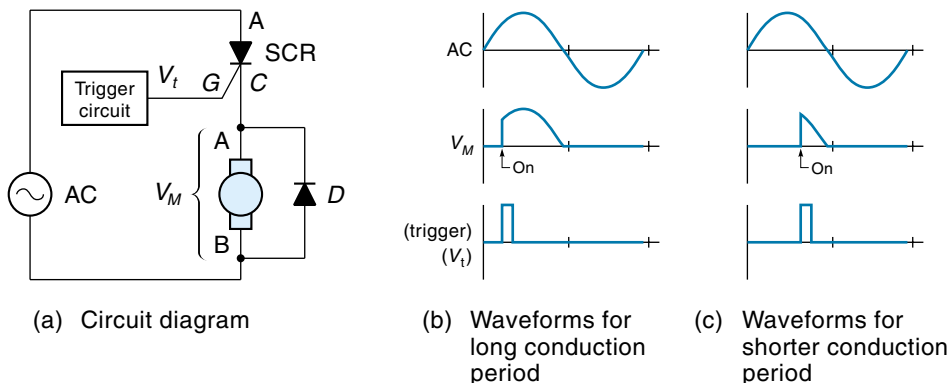
The SCR is a member of the thyristor family of semiconductor switching devices. Figure 7.31(a) shows the symbol of the SCR, which has three terminals: anode, gate, and cathode. The SCR is used as an electronic switch and can handle currents up to over 100 A. This power current ( $I_A$ ) flows from the anode to the cathode as indicated in Figure 7.31(a). The SCR is switched on by applying a low voltage to the gate ( $V_{GT}$ ), which is typically 0.6-3 V. Once turned on, it stays on, even if the gate voltage is removed. In fact, the only way to turn it off is to reduce the power current ( $I_A$ ) below a low threshold called the **holding current**, which is typically a few milliamps. Notice that the SCR symbol is similar to that of a diode, which reminds us that the SCR is a rectifier as well as a switch—that is, current can flow through it in only one direction.

Figure 7.32(a) shows the basic SCR motor control circuit. Notice that the power source is single-phase AC and that the DC motor is connected in series with the SCR. The gate of the SCR is driven by a trigger circuit that provides one pulse for each cycle of the AC. The free-wheeling diode ( $D$ ) across the motor provides an escape path for the energy stored in the motor windings when the SCR switches off.

Figures 7.32(b) and (c) show how the circuit can control motor speed. The top waveform of Figure 7.32(b) is the AC power. Notice that the motor voltage ( $V_m$ ) stays at 0 V until the trigger pulse ( $V_t$ ) turns on the SCR. Once turned on, the motor voltage equals the AC voltage for the remainder of the positive half of the cycle. During the

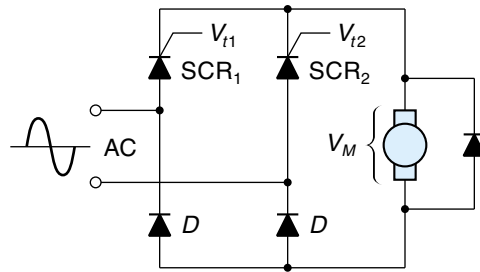
**Figure 7.32**

DC motor control from an AC source using an SCR.

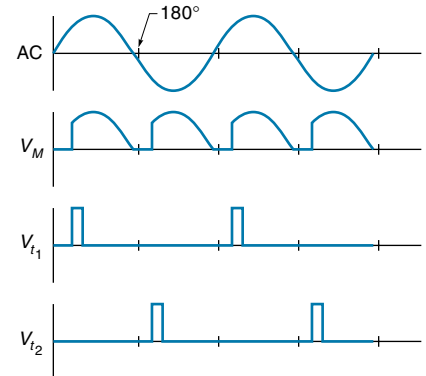


**Figure 7.33**

A full-wave SCR DC motor-control circuit using multiple SCRs.



(a) Circuit diagram



(b) Waveforms

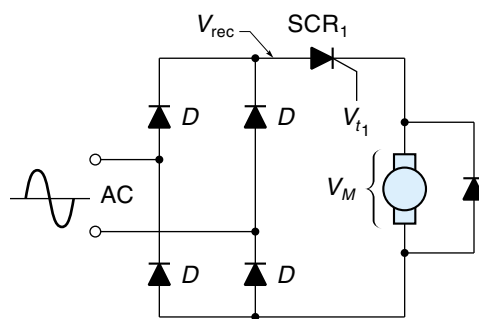
negative half of the AC cycle, the SCR remains off, and there is no power to the motor. Figure 7.32(b) shows the motor getting almost all the power this circuit can deliver, which is still only about half of the AC power available.

In Figure 7.32(c), the trigger pulse is delayed from what it was in Figure 7.32(b). Therefore, the motor is not connected to the AC power until later in the cycle, and consequently it receives even less power—that is, it runs slower. So, by controlling the delay time of the trigger pulse, we can control the speed of the motor.

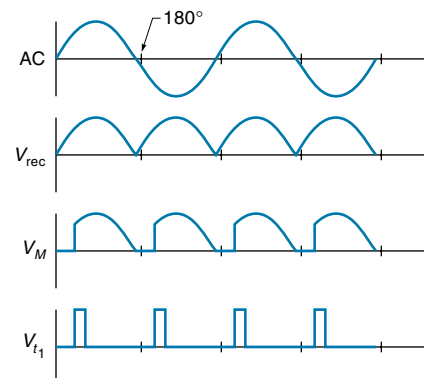
A circuit with one SCR is a half-wave rectifier, so the load gets a maximum of only half the available power. Circuits using multiple SCRs to create a full-wave rectifier overcome this problem. Figure 7.33 shows such a circuit. Here,  $SCR_1$  is triggered during the positive half of the AC cycle, and  $SCR_2$  is triggered during the negative half-cycle. The result? The motor receives two power pulses per cycle. Figure 7.34 shows another full-wave motor-control circuit. In this case, four diodes are used for the full-wave rectifier, and a single SCR controls the delay of each half cycle.

**Figure 7.34**

A full-wave SCR motor control using a full-wave bridge.



(a) Circuit diagram



(b) Waveforms

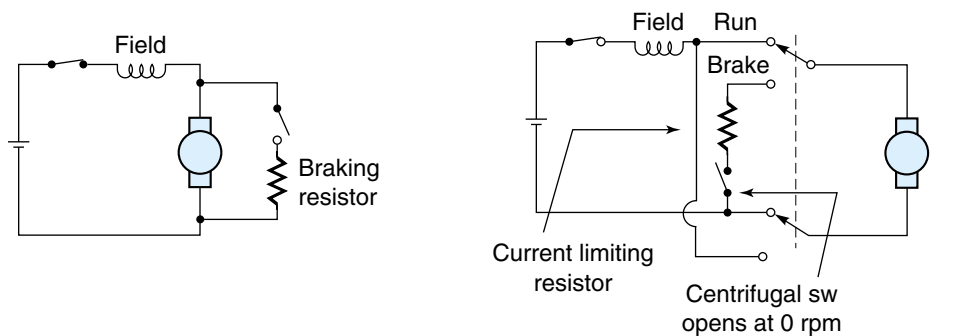
The SCR circuits described thus far are triggered somewhere in the middle of the AC positive half of the AC cycle. The resulting abrupt voltage rise generates high-frequency harmonics known as electrical “noise,” which can cause interference with other circuits, such as with radio and TV. A solution to this problem is called zero-voltage switching. With **zero-voltage switching**, the SCR is triggered on *only at the very beginning of the cycle*, when the voltage is zero anyway; consequently, there is no quick voltage change. If less than full power is desired, then, for example, only three out of four cycles would be triggered on (or some other ratio). Zero-voltage switching requires a more sophisticated trigger circuit than the phase-shift circuit discussed so far.

Electric motors have a large starting current that is many times more than the running current. For smaller motors, this may not present a problem; for larger motors (over the range of 1-2 hp), however, special **reduced voltage-starting circuits** are used. A reduced voltage-starting circuit will limit the armature current to some acceptable value when the motor starts. One way to do this is to have a resistor in series with the armature. After the motor comes up to speed, a relay is used to bypass the resistor, allowing the full line voltage to the motor.

### Braking the DC Motor

There are some applications that require the motor to stop more quickly than it would if it just coasted down. Of course, one way to stop rotation quickly is by using an external friction brake unit as discussed in Chapter 5. Other braking methods employ the motor’s existing hardware to create a resisting torque. One such method, called **dynamic braking**, uses the fact that a spinning motor becomes a generator when the power is removed. As shown in Figure 7.35(a), when the armature windings are switched to a resistor as the motor is coasting down, the “generated” current from the motor delivers power to the resistor, which dissipates the power as heat. The power to heat the resistor has to come from somewhere, and in this case it is coming from

**Figure 7.35**  
Internal braking  
circuits for the  
DC motor.



(a) Dynamic braking using a braking resistor

(b) Braking by reversing polarity (plugging) DPDT run/brake sw shown in “run” position.

the mechanical inertia of the spinning motor shaft, so the more power used up by the resistor, the sooner the shaft will come to a stop. Recall that power dissipated in a resistor is equal to the current *squared* times the resistance ( $P = I^2 \times R$ ), which indicates that the power is more influenced by a change in current than by a change in resistance. Thus the braking action can be increased by allowing more current through the resistor, which is done by making the resistance smaller. Note that this system can be used in both wound-field and PM motors, but in wound-field motors, the field coil would have to remain energized by a DC source for the generator action to work. Of course, it is possible to design a system to *use* the braking power instead of throwing it away. This is being done in electric car designs, where the power given off by the drive motor when stopping is put back in the battery; this system is known as *regenerative braking*.

Another way to brake the motor is by reversing the polarity of the armature windings and thereby causing the motor to apply a reversing torque to the load. This the kind of braking is called **plugging**. The problem is that when the voltage is reversed, it becomes the same polarity as the CEMF, so they add. This means that if you suddenly reverse the voltage on a 12-V motor, the effect is like that of putting almost 24 V across the armature. The sudden large voltage will cause a large in-rush of current, which could damage the armature. To prevent this problem, a series resistor is put in the reversing-voltage circuit [see Figure 7.35(b)]. Also, when the motor finally comes to a stop, the reversing voltage should be switched off so that the motor doesn't start to run backwards. This switching could be accomplished with a centrifugal switch similar to that shown in Figure 9.16.

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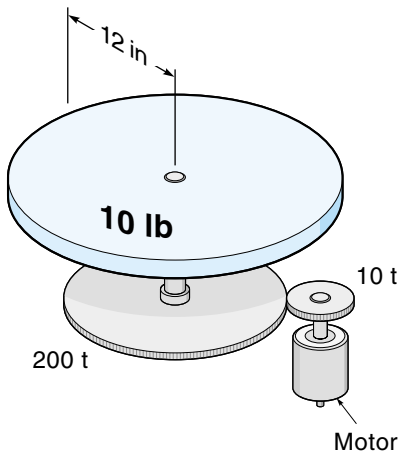
## 7.5 A COMPREHENSIVE APPLICATION USING A SMALL DC MOTOR

Example 7.7 illustrates a practical motor-application problem. The process requires you to understand the characteristics of the entire system. This example will give you an idea of the real-life factors involved in motor selection. (This example integrates some material from Chapter 5.)

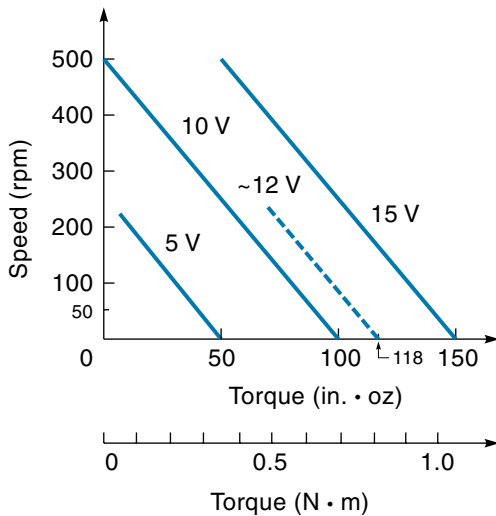
### EXAMPLE 7.7

A PM motor turns a large 24-in. diameter, 10-lb turntable through a 20 : 1 gear train [Figure 7.36(a)]. A particular requirement is that the turntable must be able to accelerate from a rest position to 90° in 0.2 s. Determine the necessary motor voltage. The torque-speed curves of the motor are given in Figure 7.36(b).

**Figure 7.36**  
Motor driving  
a turntable  
(Example 7.7).



(a) Hardware setup



(b) Motor torque–speed curves

**SOLUTION**

To solve this real-life problem, we must know how to calculate the forces required to get parts moving and understand the torque-speed characteristics of DC motors. Accelerating from 0 to 90° in 0.2 s is a response-time requirement and is determined by the system’s moment of inertia (*I*). Because the turntable is massive compared to the motor, we will only consider the turntable’s moment of inertia in the problem. First, calculate the turntable’s *I*. For a disk,

$$I = \frac{1}{2}mr^2 \tag{7.16}$$

where

*I* = moment of inertia

*m* = mass (in this case,  $m = w/g = 10 \text{ lb}/32 \text{ ft/s}^2$ )

*r* = radius (in this case, 12 in., or 1 ft)

After plugging in the numbers, we get

$$I = \frac{1}{2} \frac{10 \text{ lb} \text{ s}^2}{32 \text{ ft}} \cdot 1 \text{ ft}^2$$

$$= 0.156 \text{ lb} \cdot \text{s}^2 \cdot \text{ft}$$

Converting  $I$  to  $\text{in.} \cdot \text{oz} \cdot \text{s}^2$  gives us

$$I = 0.156 \text{ lb} \cdot \text{s}^2 \cdot \text{ft} \times \frac{16 \text{ oz}}{\text{lb}} \times \frac{12 \text{ in.}}{\text{ft}} = 30 \text{ in.} \cdot \text{oz} \cdot \text{s}^2$$

which is the moment of inertia for the turntable.

Second, we need to know the acceleration required. From Chapter 5, recall the equation that relates position to acceleration:

$$\theta = \frac{1}{2}\alpha t^2 \quad (7.17)$$

where

$\theta$  = angle in radians

$\alpha$  = angular acceleration

$t$  = elapsed time since the object was at rest

Solving for acceleration, we find that

$$\alpha = \frac{2\theta}{t^2} = \frac{2(90^\circ)}{(0.2 \text{ s})^2} \times \frac{\pi \text{ rad}}{180^\circ} = 79 \text{ rad/s}^2$$

Now we need to determine the motor torque required to cause this acceleration. The basic equation of angular motion is

$$T = I\alpha \quad (7.18)$$

where  $T$  is the motor torque. But remember that the motor is connected to the turntable through a 20 : 1 gear ratio. Thus the turntable moment of inertia reflected back to the motor is only 1/20 squared as big, and the motor must accelerate 20 times faster than the turntable, as shown in the following equation:

$$T = I\alpha = \underbrace{\frac{(30 \text{ in.} \cdot \text{oz} \cdot \text{s}^2)}{20^2}}_{\substack{I \text{ of table} \\ \text{(as reflected} \\ \text{to motor)}}} \times \underbrace{\frac{(79 \times 20)}{\text{s}^2}}_{\substack{\alpha \text{ required} \\ \text{of motor}}} = 118 \text{ in.} \cdot \text{oz}$$

This result shows that 118 in. · oz of motor torque are required to accelerate the turntable. Looking along the  $x$ -axis of the torque-speed curves [Figure 7.36(b)] we see that for a stall torque of 118 in. · oz, the motor must get approximately 12 V. The answer to the original question seems to be as follows: The motor voltage to spin the turntable 90° in 0.2 s equals 12 V.

**Final Comment**

There is a problem, however, with this conclusion because it is based on the assumption that the torque is a *constant* 118 in. · oz. It is true that when the motor is at a dead stop, 12 V will cause a torque of 118 in. · oz, but by the time the turntable has moved 90°, the motor is turning at 50 rps and the torque is down to 110 in. · oz. In this case, the difference between 118 and 110 in. · oz is not that much and may (with minor adjustments) be considered an acceptable solution. Getting the precise answer to this problem would require either mathematics beyond the scope of this text or empirical testing.

**EXAMPLE 7.7 (Repeated with SI Units)**

A PM motor turns a large 60-cm diameter, 4.5-kg turntable through a 20 : 1 gear train [Figure 7.36(a)]. A particular requirement is that the turntable must be able to accelerate from a rest position to 90° in 0.2 s. Determine the necessary motor voltage. The torque-speed curves of the motor are given in Figure 7.36(b).

**SOLUTION**

Accelerating from 0 to 90° in 0.2 s is a response-time requirement and is determined by the system's moment of inertia ( $I$ ). Because the turntable is massive compared to the motor, we will only consider the turntable's moment of inertia in the problem. First, calculate the turntable's  $I$ . For a disk,

$$I = \frac{1}{2}mr^2 \quad (7.16)$$

where

$I$  = moment of inertia

$m$  = mass (in this case, 4.5 kg)

$r$  = radius (in this case, 30 cm)

After plugging in the numbers, we get

$$\begin{aligned} I &= \frac{1}{2} 4.5 \text{ kg } 30 \text{ cm}^2 \\ &= 2025 \text{ kg} \cdot \text{cm}^2 = 0.2025 \text{ kg} \cdot \text{m}^2 \end{aligned}$$

which is the moment of inertia for the turntable.

Second, we need to know the acceleration required:

$$\theta = \frac{1}{2}at^2 \quad (7.17)$$

where

$\theta$  = angle in radians

$\alpha$  = angular acceleration

$t$  = elapsed time since the object was at rest

Solving for acceleration, we find that

$$\alpha = \frac{2\theta}{t^2} = \frac{2(90^\circ)}{(0.2s)^2} \times \frac{\pi \text{ rad}}{180^\circ} = 79 \text{ rad/s}^2$$

Now we need to determine the motor torque required to cause this acceleration. The basic equation of angular motion is

$$T = I\alpha \quad (7.18)$$

where  $T$  is the motor torque. But remember that the motor is connected to the turntable through a 20 : 1 gear ratio. Thus the turntable moment of inertia reflected back to the motor is only 1/20 squared as big, and the motor must accelerate 20 times faster than the turntable, as shown in the following equation:

$$T = I\alpha = \underbrace{\frac{(0.2025 \text{ kg} \cdot \text{m}^2)}{20^2}}_{\substack{I \text{ of table} \\ \text{(as reflected} \\ \text{to motor)}}} \times \underbrace{\frac{(79 \times 20)}{s^2}}_{\substack{\alpha \text{ required} \\ \text{of motor}}} = 0.8 \text{ N} \cdot \text{m}$$

[*Note:* The relationship  $1 \text{ kg} = 1\text{N}/(1 \text{ m/s}^2)$  was used to get units to cancel.]

This result shows that 0.8 N · m of motor torque is required to accelerate the turntable. Looking along the  $x$ -axis of the torque-speed curves [Figure 7.36(b)] we see that for a stall torque of 0.8 N · m, the motor must get approximately 12 V. The answer to the original question seems to be as follows: The motor voltage to spin the turntable 90° in 0.2 s equals 12 V. (However, see “Final Comment” on page 333.)



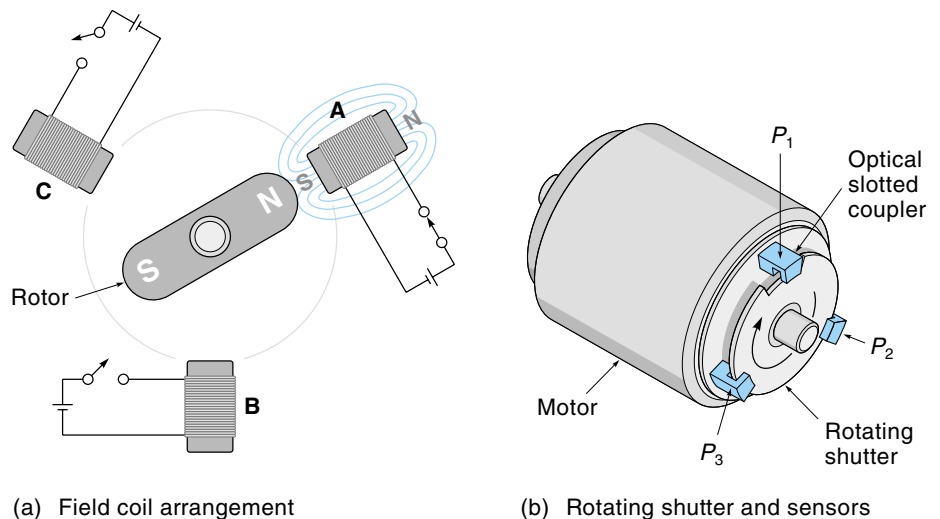
## 7.6 BRUSHLESS DC MOTORS

The weak point in the mechanical design of the DC motor is the brushes rubbing against the rotating commutator (to get current into the armature). Brushes wear out, get dirty, cause dust, and are electrically noisy. The **brushless DC motor (BLDC)** operates without brushes by taking advantage of modern electronic switching techniques. Although this adds some complexity, the result is a motor that is extremely reliable, very efficient, and easily controlled—all very desirable qualities. The BLDC is becoming increasingly popular, particularly in those cases where the motor must be operated from a DC source such as a battery.

Figure 7.37(a) shows a diagram of a three-phase BLDC. The armature (called the **rotor**) is a permanent magnet, and it is surrounded by three field coils. Each field coil can be switched on and off independently. When a coil is on, such as coil A in Figure 7.37(a), the north pole of the rotor magnet is attracted to that coil. By switching the coils on and off in sequence (A, B, C), the rotor is “dragged” around clockwise—that is, the field has rotated electronically.

BLDCs have much in common with stepper motors, which are discussed in detail in Chapter 8. The major difference between these two types of motors is that the BLDC is used as a source of rotary power, like a regular electric motor, whereas the stepper motor is used when it is necessary to step to precise positions and then stop. Unlike the stepper motor, the BLDC has a built-in sensor system to direct the switching from one field coil to the next. Figure 7.37(b) shows the three-phase BLDC with three optical slotted couplers and a rotating shutter (Hall-effect sensors can also be used for this application). These position sensors control the field windings. When the shutter is open for sensor  $P_1$  as shown, field coil A [Figure 7.37(a)] is energized. When the rotor actually

**Figure 7.37**  
A three-phase brushless  
DC motor.



gets to field coil A, sensor  $P_1$  is turned off and  $P_2$  is turned on, energizing field coil B and pulling the rotor on around to coil B, and so on. In this manner, the rotor is made to rotate with no electrical connection between the rotor and the field housing.

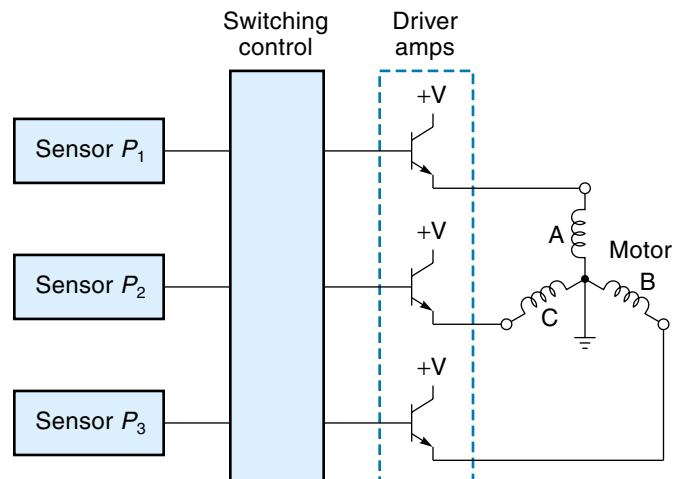
Figure 7.38 shows a schematic of a generalized three-phase BLDC. The three position sensors connect to the control circuitry. In the simplest case, such as described in the preceding paragraph, these signals are passed directly on to solid-state switches that drive the motor coils. A more sophisticated motor-control system would provide for the motor to reverse direction (by reversing the sequencing) and would control the speed by using PWM techniques.

Since its introduction in 1962, the BLDC motor has been undergoing continuous improvement. In particular, the availability of improved magnet materials, better semiconductors for controlling the power, and application-specific integrated circuits (ASIC) have helped boost the BLDC to higher performance levels. BLDCs are available from fractional-hp to 600 hp models. The larger models are powered from rectified AC using circuits similar to that shown in Figure 7.33. The DC is then switched rapidly on and off with power MOSFETs or bipolar transistors to create PWM drive to the motor coils.

BLDC motors exhibit excellent speed control. In fact, some models come with a built-in tachometer that feeds back to the control unit, allowing a speed regulation of 0% (perfect). When used in a variable-speed motion-control system, BLDCs can vary their speed in the range of 100:1. However, unlike a brushed DC motor, the BLDC has a minimum operating speed (around 300 rpm) below which the individual power pulses can be felt (called *cogging*).

Besides being more reliable, modern BLDC motors have performance advantages over brushed DC motors and even AC induction motors (AC motors are covered in Chapter 9). Specifically, BLDC motors have higher power efficiency (they use less

**Figure 7.38**  
Schematic of a three-phase BLDC.



power for the same horsepower) and are smaller and lighter than other types of motors with the same horsepower. It seems reasonable to conclude that the BLDC motor will continue to improve and be adopted in more and more applications.

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## SUMMARY

The electric motor, the most common type of actuator, is based on the principle that a current-carrying conductor will experience a physical force when in the presence of a magnetic field. The DC motor consists of a rotating armature and a stationary magnetic field. The current in the armature, which must come through brushes, causes the rotating forces. The stationary magnetic field is provided by either electromagnets (in which case, it is called a wound-field motor) or by permanent magnets.

There are three types of wound field motors. The series motor has the armature and field windings connected in series. This type of motor is characterized by a high starting torque and a high no-load speed but poor speed regulation (the speed changes considerably if the load changes). The shunt motor has the armature and field windings connected in parallel. This type of motor has much better speed regulation than does the series motor. The compound motor has both series and shunt-type field windings and combines the good characteristics of both the series and shunt motors.

Permanent magnet (PM) motors use permanent magnets to provide the stationary magnetic field. This results in a very linear torque-speed curve, which makes it easy to calculate the motor speed for various load conditions and thus attractive for control system applications.

There are two ways to control the speed of a DC motor: (1) Analog drive uses linear amplifiers to provide a varying DC voltage to the motor. Although simple and direct, this method is very power-inefficient and is usually used only with smaller motors. (2) Pulse-width modulation (PWM), a more efficient method, provides the motor with constant voltage pulses of varying widths: the wider the pulse, the more energy transferred to the motor. For larger DC motors, SCRs power the motor with pulses taken directly from the AC waveform. This is a form of PWM, and it eliminates the need for a large DC-power supply.

The newest type of DC motor is the brushless DC motor (BLDC). This motor uses permanent magnets instead of coils in the armature (called the rotor) and so does not need brushes. The field coils are switched on and off in a rotating sequence that pulls the rotor around. BLDCs have built-in sensors that direct when the individual field coils are to be switched on and off.

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## GLOSSARY

**analog drive** A method of controlling an electric motor's speed by varying the DC supply voltage.

**armature** The part of a motor that responds to the magnetic field; typically, the armature is the rotating assembly of an electric motor.

**BLDC** See **brushless DC motor**.

**brush** A stationary conductive rod that rubs on the rotating commutator and conducts current into the armature windings.

**brushless DC motor (BLDC)** The newest type of DC motor that does not use a commutator or brushes; instead, the field windings are turned on and off in sequence, and the permanent magnet rotor (armature) is pulled around by the apparently rotating magnetic field.

**CEMF** See **counter-EMF**.

**commutator** The part of an armature that makes contact with the brushes.

**compound motor** A motor that has both shunt and series field windings.

**counter-EMF (CEMF)** A voltage that is generated inside an electric motor when running under its own power; the polarity of the CEMF is always opposite to the applied voltage.

**cumulative compound motor** A compound motor where the magnetic fields of the shunt and series fields aid each other.

**differential compound motor** A compound motor where the magnetic fields of the shunt and series windings oppose each other.

**duty cycle** The percentage of cycle time that the PWM pulse is high—that is, a true square wave has a duty cycle of 50%.

**dynamic braking** A method of braking the motor by switching the armature windings to a resistor. The coasting motor generates current that is dissipated in the resistor, thus removing energy from the system.

**field** The part of an electric motor that provides a magnetic field; typically, it is the stationary part.

**field winding** A stationary electromagnet used to provide the magnetic field needed by the armature.

**gearhead** In smaller motors, a gear train attached or built-in to the motor assembly, which effectively gives the motor more torque at less rpm.

**holding current** Once the SCR has been turned on, the small current (to the SCR) necessary to keep the SCR in the conduction state.

*/ See moment of inertia.*

**load torque** The torque required of the motor to rotate the load.

**moment of inertia ( $I$ )** A quantity based on mass and shape of a rotating part; the larger  $I$  is, the more torque it takes to change the rpm of the part.

**no-load speed** The speed of a motor when there is no external load on it; it will always be the maximum speed (for a particular voltage).

**permanent-magnet (PM) motors** A motor that uses permanent magnets to provide the magnetic field. The PM motor has a linear torque-speed relationship, making it desirable for control applications.

**plugging** A method of braking the motor by reversing the polarity of the armature windings, causing the motor to apply a reversing torque to the load.

**PM motor** *See* **permanent-magnet motor.**

**pulse-width modulation (PWM)** A method of controlling an electric motor's speed by providing pulses that are at a constant DC voltage. The width of the pulses is varied to control the speed.

**PWM** *See* **pulse-width modulation.**

**rated speed** The speed of a motor when producing its rated horsepower.

**reduced voltage-starting circuit** A special circuit that limits the motor current during start-up.

**rotor** The rotating part of a motor; if the field poles are stationary (as they usually are), then the rotor is known as the armature.

**SCR** *See* **silicon-controlled rectifier.**

**series-wound motor** A motor that has the field windings connected in series with the armature; the series-wound motor has a high starting torque and a high no-load speed.

**shunt-wound motor** A motor that has the field windings connected in parallel with the armature windings; the shunt-wound motor has a measure of natural speed regulation—that is, it tends to maintain a certain speed despite load changes.

**silicon-controlled rectifier (SCR)** A semiconductor device that provides speed control to a DC motor from an AC-power source (without the need of a power supply).

**speed regulation** In general, a motor's ability to maintain its speed under different loads; specifically, a percentage based on no-load speed and full-load speed.

**stall torque** The torque of the motor when the shaft is prevented from rotating; it will always be the maximum torque (for a particular voltage).

**torque** In general, the measure of the motor's strength in providing a twisting force; specifically, the product of a tangential force times the radius.

**torque-speed curve** A graph of a motor's torque versus speed; can be used to predict the motor's speed under various load conditions.

**zero-voltage switch** A technique of switching on an AC device (such as an SCR) only at the beginning of the cycle when the voltage is zero anyway. This system eliminates sharp voltage rises at turn-on and reduces the generation of high-frequency interference.

## EXERCISES

### Section 7.1

1. What is CEMF? How and why does it affect electric motor performance?
2. A DC motor is running with a line voltage of 90 V, a CEMF of 80 V, and an armature resistance of  $3\ \Omega$ . Find the armature current.
3. A motor generates a CEMF at a rate of 0.4 V/100 rpm and has an armature resistance of  $20\ \Omega$ . Ten volts are applied to the (stopped) motor. Find the armature current just after the power is applied and when the speed is 1500 rpm.
4. A motor generates a CEMF at a rate of 0.5 V/100 rpm and has an armature resistance of  $15\ \Omega$ . Twelve volts are applied to the (stopped) motor. Find the armature current just after the power is applied and when the speed is 1500 rpm.

### Section 7.2

5. What are the distinguishing characteristics of the series-wound DC motor?
6. Explain *stall torque* and *no-load speed*.
7. What are the distinguishing characteristics of the shunt-wound motor?
8. When 24 V are applied to a motor with no load, the speed levels off at 2300 rpm. When the load is connected, the speed drops to 2050 rpm. Calculate the speed regulation of this motor.
9. When 90 V are applied to a motor with no load, the speed levels off at 2000 rpm. When the load is connected, the speed drops to 1750 rpm. Calculate the speed regulation of this motor.
10. Select a motor from the list of Figure 7.4 that meets the following requirements: Load torque = 3 ft · lb, and speed = 1750 rpm.

### Section 7.3

11. Why is the PM motor particularly suited to control system applications?
12. A PM motor supplied with 10 V has the torque-speed curves of Figure 7.10.
  - a. Find the stall torque.
  - b. Find the no-load speed.
  - c. Find the rpm of the motor if it is lifting a 4-oz load by a string over a 3-in. radius pulley.
  - d. A 3 : 1 step-down gear pass is inserted between the motor and pulley shaft (of part c). Sketch the setup and determine the motor rpm.

13. A PM motor has the torque-speed curves of Figure 7.10. You want to use the motor in an application that requires a speed of 450 rpm and a load of 11 in. · oz. What voltage must be supplied to the motor?
14. A motor has the torque-speed curves of Figure 7.10. If the supply voltage is 12 V, what is the largest torque the motor can deliver and still maintain a speed of 600 rpm?
15. For the motor in Figure 7.11(a), draw the torque-speed curve for model 22-45-16.
16. For model 22-45-12 in Figure 7.11(a), find the approximate motor torque if the applied voltage is 15 V and speed is 5000 rpm.

### Section 7.4

17. Comment on the power efficiency of an analog-drive motor-control system that uses a class A power amplifier.
18. An analog-drive circuit uses a power Darlington transistor. The supply voltage is 10 V; at 300 rpm, the motor is drawing 500 mA with a voltage of 3 V. How much power is the transistor dissipating?
19. An analog-drive circuit uses a power Darlington transistor. The supply voltage is 12 V; at 400 rpm, the motor is drawing 1 A with a voltage of 7 V. How much power is the transistor dissipating?
20. Explain the principle of PWM for motor-speed control.
21. What duty cycle would you specify to make a 2000-rpm, 12-V motor run at 1500 rpm? (Assume 12-V pulses and no PWM losses.)
22. What duty cycle would you specify to make a 1500-rpm, 12-V motor run at 1000 rpm? (Assume 12-V pulses and no PWM losses.)
23. What are the advantages of PWM over analog drive for motor control?
24. Explain how an SCR is turned on and off.
25. Sketch the voltage waveform of a full-wave SCR-driven motor for the following conditions:
  - a. SCR starts conducting at  $30^\circ$ .
  - b. SCR starts conducting at  $120^\circ$ .

### Section 7.5

26. A rotating antenna has a moment of inertia ( $I$ ) of  $0.8 \text{ lb} \cdot \text{ft} \cdot \text{s}^2$ . It is driven through a 10 : 1 gearbox by a DC motor (motor turns 10X faster). The antenna must be able to rotate  $180^\circ$  from rest in 1 s. The motor characteristics are in Figure 7.36. Find the motor voltage required.
27. A rotating antenna has a moment of inertia ( $I$ ) of  $1 \text{ kg} \cdot \text{m}^2$ . It is driven through a 10 : 1 gearbox by a DC motor. The antenna must be able to rotate  $180^\circ$  from rest in 1 s. The motor characteristics are in Figure 7.36. Find the motor voltage required.

### Section 7.6

28. Explain how a brushless DC motor works without brushes.
29. What are the advantages of a BDCM over a regular DC motor?





# CHAPTER 8

## Stepper Motors

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### OBJECTIVES

After studying this chapter, you should be able to:

- Explain what a stepper motor is, how it is different from a “regular” motor, and the applications it is used in.
- Understand the basic parts and operation of the three kinds of stepper motors: permanent magnet, variable reluctance, and hybrid.
- Differentiate between two-phase, three-phase, and four-phase stepper motors.
- Understand the different operational modes—single-step versus slew, single- and dual-phase excitation, half-step, and microstepping.
- Calculate the final position of a stepper motor, given the sequence of drive pulses.
- Explain the operation of stepper motor driver circuits.

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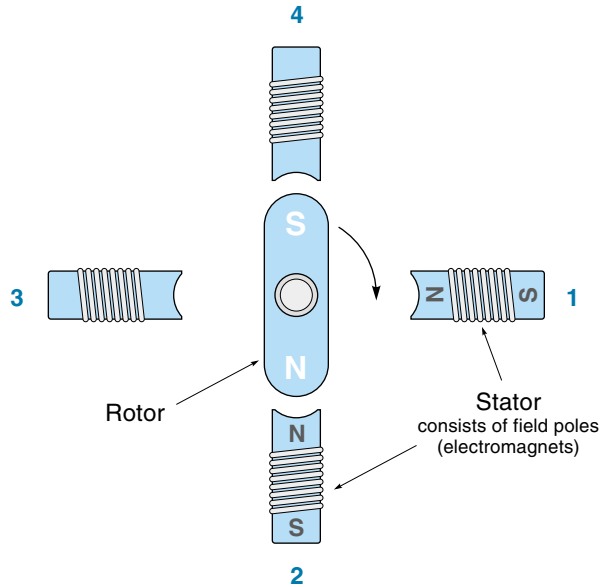
### INTRODUCTION

A **stepper motor** is a unique type of DC motor that rotates in fixed steps of a certain number of degrees. Step size can range from 0.9 to 90°. Figure 8.1 illustrates a basic stepper motor, which consists of a **rotor** and **stator**. In this case, the rotor is a permanent magnet, and the stator is made up of electromagnets (field poles). The rotor will move (or step) to align itself with an energized field magnet. If the field magnets are energized one after the other around the circle, the rotor can be made to move in a complete circle.

Stepper motors are particularly useful in control applications because the controller can know the exact position of the motor shaft without the need of position sensors. This is done by simply counting the number of steps taken from a known reference position. Step size is determined by the number of rotor and stator poles, and there is no **cumulative error** (the angle error does not increase, regardless of the number of steps taken). In fact, most stepper motor systems operate open-loop—that is, the controller sends the motor a determined number of step commands and assumes the motor

**Figure 8.1**

A PM 90 stepper motor.



goes to the right place. A common example is the positioning of the read/write head in a floppy disk drive.

Steppers have inherently low velocity and therefore are frequently used without gear reductions. A typical unit driven at 500 pulses/second rotates at only 150 rpm. Stepper motors can easily be controlled to turn at 1 rpm or less with complete accuracy.

There are three types of stepper motors: permanent magnet, variable reluctance, and hybrid. All types perform the same basic function, but some differences among them may be important in some applications.

## 8.1 PERMANENT-MAGNET STEPPER MOTORS

The **permanent-magnet (PM) stepper motor** uses a permanent magnet for the rotor. Figure 8.1 shows a simple PM stepper motor. The field consists of four poles (electromagnets). The motor works in the following manner: Assume the rotor is in the position shown with the south end up. When field coil 1 is energized, the south end of the rotor is attracted to coil 1 and moves toward it. Then field coil 1 is deenergized, and coil 2 is energized. The rotor pulls itself into alignment with coil 2. Thus, the rotor turns in 90° steps for each successive excitation of the field coils. The motor can be made to reverse by inverting the sequence.

One desirable property of the PM stepper motor is that the rotor will tend to align up with a field pole even when no power is applied because the PM rotor will be attracted to the closest iron pole. You can feel this “magnetic tug” if you rotate the motor

by hand; it is called the detent torque, or residual torque. The detent torque is a desirable property in many applications because it tends to hold the motor in the last position it was stepped to, even when all power is removed.

As mentioned earlier, one big advantage of the stepper motor is that it can be used open-loop—that is, by keeping track of the number of steps taken from a known point, the exact shaft position is always known. Example 8.1 demonstrates this.

### EXAMPLE 8.1

A  $15^\circ$ /step stepper motor is given 64 steps CW (clockwise) and 12 steps CCW (counterclockwise). Assuming it started at  $0^\circ$ , find the final position.

### SOLUTION

After completing 64 steps CW and 12 steps CCW, the motor has ended up 52 steps CW ( $64 - 12 = 52$ ). Because there are 24  $15^\circ$ -steps per revolution ( $360^\circ/15^\circ = 24$ ),

$$\begin{aligned}\frac{52 \text{ steps}}{24 \text{ steps/rev}} &= 2\frac{1}{6} = 2 \text{ rev} + \frac{1}{6} \text{ rev} \\ &= 2 \text{ rev} + \frac{360^\circ}{6} \\ &= 2 \text{ rev} + 60^\circ\end{aligned}$$

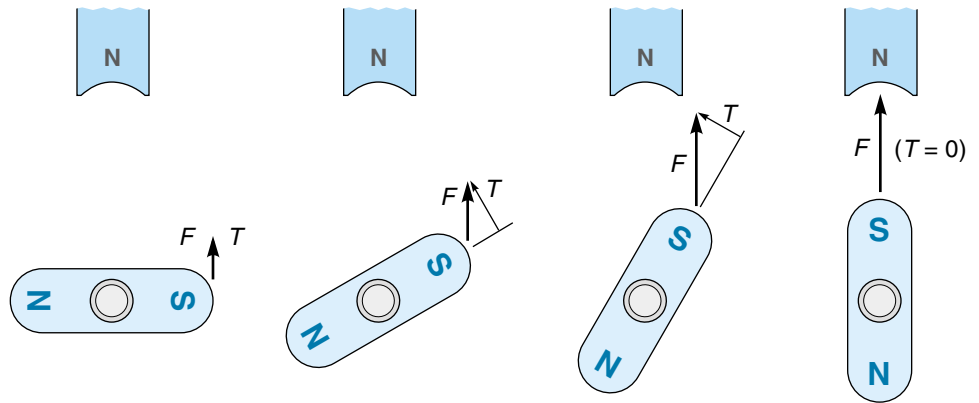
Therefore, the motor has made two complete revolutions and is now sitting at  $60^\circ$  CW from where it started.

## Effect of Load on Stepper Motors

For the open-loop concept to work, the motor must actually step once each time it's commanded to. If the load is too great, the motor may not have enough torque to make the step. In such a case, the rotor would probably rotate a little when the step pulse was applied but then fall back to its original position. This is called **stalling**. If feedback is not used, the controller has no way of knowing a step was missed.

Within each step, the torque developed by the stepper motor is dependent on the shaft angle. In fact, the torque on the rotor is actually zero when it is exactly aligned with an energized field coil. Figure 8.2 illustrates how the motor can only provide torque when the rotor is *not* aligned. The first frame of Figure 8.2 shows a rotor pole approaching an energized field pole. The actual force of attraction is between the south (S) end of the rotor and the north (N) end of the field pole. As the rotor pole approaches the field pole, the attraction force ( $F$ ) gets stronger but the torque component ( $T$ ) gets weaker. When the rotor is pointing directly at the field pole (last frame in Figure 8.2),

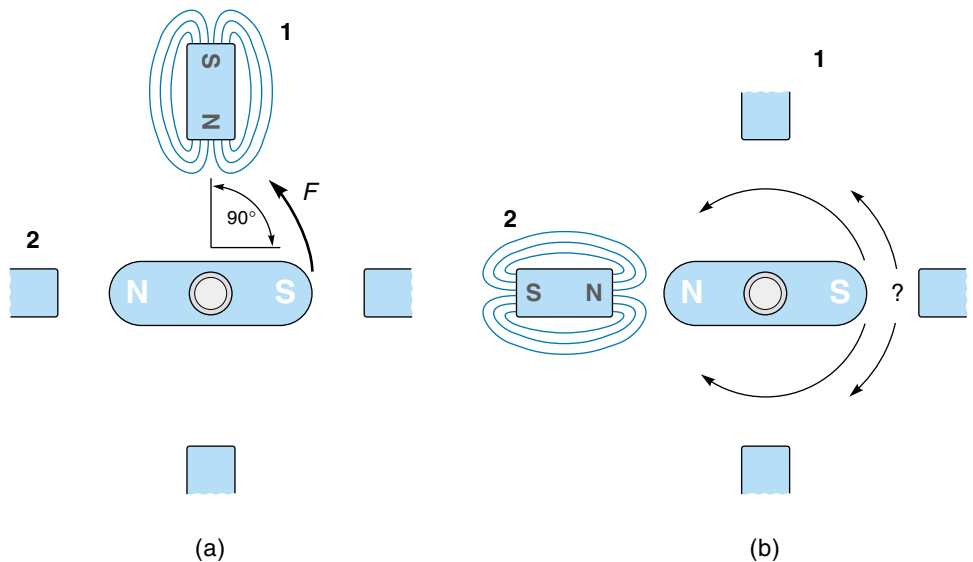
**Figure 8.2**  
Torque goes to zero as the rotor aligns with the field pole.



the torque component is zero. In practice, this means that the rotor may come to a stop before it is completely aligned with the energized field pole, at the point where the diminishing step torque just equals the load torque.

For the simple motor under discussion (Figure 8.1), the maximum torque occurs when the rotor is somewhere around  $45^\circ$  away from the field pole (second frame of Figure 8.2.) If the load exceeds this maximum motor torque, the rotor will quickly slip to  $90^\circ$  behind, and this might cause the motor to take a step backward instead of forward. Consider the situation in Figure 8.3(a) where the rotor has been stepping CCW and we have allowed it to lag a full step behind the energized pole (currently, pole 1).

**Figure 8.3**  
Illustrating what would happen if the rotor were allowed to lag a full step behind the field poles.



The next pole to be energized in the CCW sequence is pole 2 [Figure 8.3(b)]. The first problem here is that the rotor is pointing directly away from pole 2, so there will be little or no torque exerted. The second problem is that, in this balanced condition, the rotor will be equally attracted in either direction and we cannot reliably predict if it will turn CW or CCW.

For proper operation, *the rotor lag must not be allowed exceed one-half the step size*, which would be  $45^\circ$  for the motor illustrated in Figure 8.3. This solves the preceding problems—namely, the motor will always turn in the direction it's supposed to, and it will not stall. (Recall that stalling occurs when the motor is too weak to take a step.) In practical terms, the *dynamic torque*, which is the torque available when the motor is running, may only be about half of the maximum *holding torque* (the torque required to displace the rotor when stopped). There is an exception to this rule: When the rotor is stepping rapidly (called *slewing*), the inertia can be counted on to keep the rotor going in the right direction. Slewing is discussed in the next section.

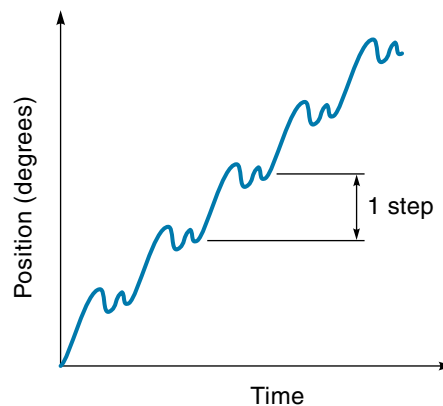
## Modes of Operation

The stepper motor has two modes of operation: single step and slew. In the **single-step mode** or bidirectional mode, the frequency of the steps is slow enough to allow the rotor to (almost) come to a stop between steps. Figure 8.4 shows a graph of position versus time for single-step operation. For each step, the motor advances a certain angle and then stops. If the motor is only lightly loaded, overshoot and oscillations may occur at the end of each step as shown in the figure.

The big advantage of single-step operation is that each step is completely independent from every other step—that is, the motor can come to a dead stop or even reverse direction at any time. Therefore, the controller has complete and instantaneous control of the motor's operation. Also, there is a high certainty that the controller will not lose count (and hence motor position) because each step is so well defined. The disadvantage

**Figure 8.4**

Position versus time for the single-step mode.



of single-step mode is that the motion is slow and “choppy.” A typical single-step rate is 5 steps/second which translates to 12.5 rpm for a  $15^\circ$ /step motor.

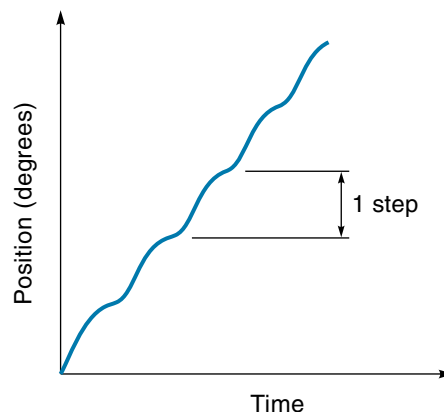
In the **slew mode**, or unidirectional mode, the frequency of the steps is high enough that the rotor does not have time to come to a stop. This mode approximates the operation of a regular electric motor—that is, the rotor is always experiencing a torque and rotates in a smoother, continuous fashion. Figure 8.5 shows a graph of position versus time for the slew mode. Although the individual steps can still be discerned, the motion is much less choppy than in single-step mode.

A stepper motor in the slew mode cannot stop or reverse direction instantaneously. If attempted, the rotational inertia of the motor would most likely carry the rotor ahead a few steps before it came to rest. The step-count integrity would be lost. It is possible to maintain the step count in the slew mode by slowly ramping up the velocity from the single-step mode and then ramping down at the end of the slew. This means the controller must know ahead of time how far the motor must go. Typically, the slew mode is used to get the motor position in the “ballpark,” and then the fine adjustments can be made with single steps. Slewing moves the motor faster but increases the chances of losing the step count.

Figure 8.6 shows the torque-speed curves for both the single-step and slew modes. The first observation is that available load torque diminishes as the stepping rate rises (this is true of all DC motors). Also, for the single-step mode, the price paid for the ability to stop or reverse instantaneously is less torque and speed. Looking along the  $x$ -axis, notice three different kinds of torques. The **detent torque** is the torque required to overcome the force of the permanent magnets (when the power is off). It is the little tugs you feel if you manually rotate the unpowered motor. The **dynamic torque**, which is the maximum running torque, is obtained when the rotor is lagging behind the field poles by half a step. The highest stall torque shown in Figure 8.6 is called the **holding torque** and results when the motor is completely stopped but with the last pole still energized. This is really a detent type of torque because it represents the amount of external torque needed to rotate the motor “against its wishes.”

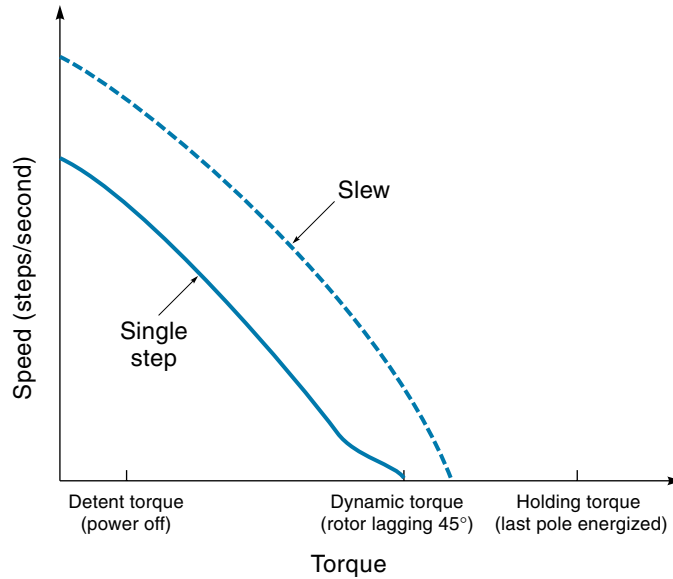
**Figure 8.5**

Position versus time for the slew mode.



**Figure 8.6**

Torque–speed curves for single-step and slew modes.

**EXAMPLE 8.2**

A stepper motor has the following properties:

Holding torque: 50 in. · oz

Dynamic torque: 30 in. · oz

Detent torque: 5 in. · oz

The stepper motor will be used to rotate a 1-in. diameter printer platen (Figure 8.7). The force required to pull the paper through the printer is estimated to not exceed 40 oz. The static weight of the paper on the platen (when the printer is off) is 12 oz. Will this stepper motor do the job?

**SOLUTION**

The torque required to rotate the platen during printing can be calculated as follows:

$$\text{Torque} = \text{force} \times \text{radius} = 40 \text{ oz} \times 0.5 \text{ in.} = 20 \text{ in.} \cdot \text{oz}$$

Therefore, the motor, with 30 in. · oz of dynamic torque, will be strong enough to advance the paper.

The torque on the platen from just the weight of the paper is calculated as follows:

$$\text{Torque} = \text{force} \times \text{radius} = 12 \text{ oz} \times 0.5 \text{ in.} = 6 \text{ in.} \cdot \text{oz}$$

When the printer is on, the powered holding torque of 50 in. · oz is more than enough to support the paper. However, when the printer is off, the weight of the paper exceeds the detent torque of 5 in. · oz, and the platen (and motor) would spin backward. Therefore, we conclude that this motor is not acceptable for the job (unless some provision such as a ratchet or brake is used to prevent back spinning).

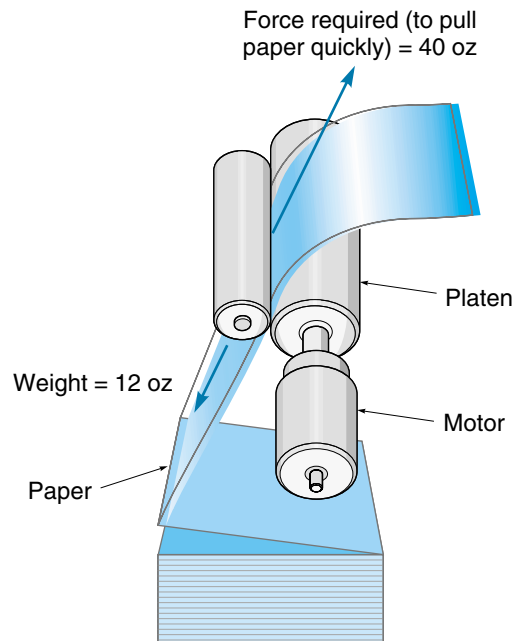
## Excitation Modes for PM Stepper Motors

Stepper motors come with a variety of winding and rotor combinations. In addition, there are different ways to sequence energy to the field coils. All these factors determine the size of each step. **Phase** refers to the number of separate winding circuits. There are two-, three-, and four-phase steppers, which are discussed next.

### *Two-Phase (Bipolar) Stepper Motors*

The **two-phase (bipolar) stepper motor** has only two circuits but actually consists of four field poles. Figure 8.8(a) shows the motor symbol, and Figure 8.8(b) shows how it is wired internally. In Figure 8.8(b), circuit AB consists of two opposing poles such that when voltage is applied (+A –B), the top pole will present a north end to the rotor and the bottom pole will present a south end. The rotor would tend to align itself vertically (position 1) with its south pole up (because, of course, opposite magnetic poles attract).

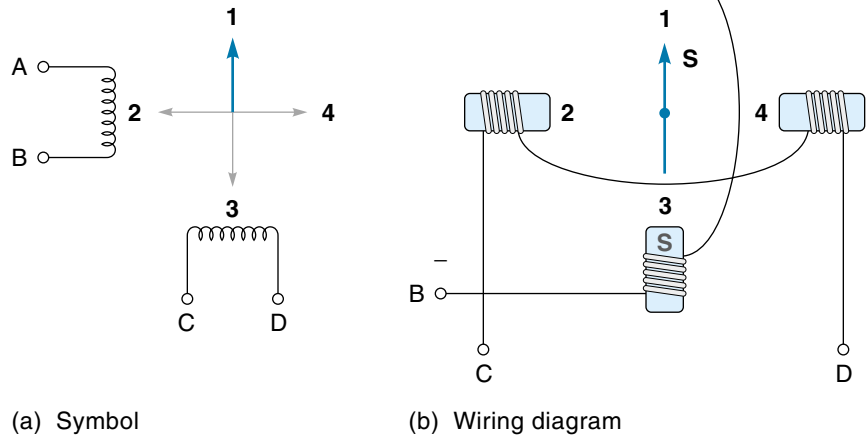
**Figure 8.7**  
A stepper motor driving a printer platen (Example 8.2).





**Figure 8.8**

A two-phase (bipolar) stepper motor.



The simplest way to step this motor is to alternately energize either AB or CD in such a way as to pull the rotor from pole to pole. If the rotor is to turn CCW from position 1, then circuit CD must be energized with polarity C+ D-. This would pull the rotor to position 2. Next, circuit AB is energized again, but this time the polarity is reversed (-A +B), causing the bottom pole to present a north end to the rotor, thereby pulling it to position 3. The term **bipolar** applies to this motor because the current is sometimes reversed. The voltage sequence needed to rotate the motor one full turn is shown below. Reading from top to bottom gives the sequence for turning CCW, reading from bottom up gives the CW sequence:

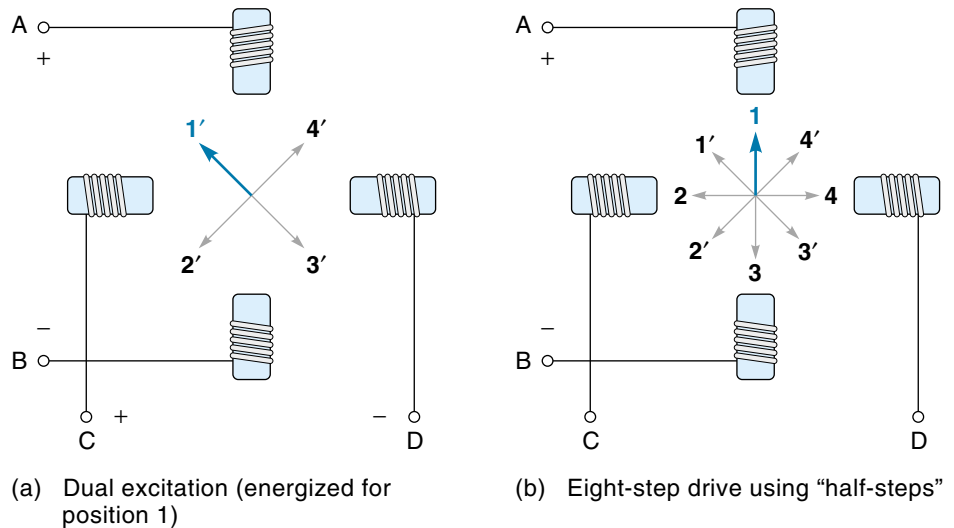
Circuit	Position	
A+ B-	1	
C+ D-	2	(excitation for Figure 8.8)
A- B+	3	
C- D+	4	

Another way to operate the two-phase stepper is to energize both circuits at the same time. In this mode, the rotor is attracted to two adjacent poles and assumes a position in between. Figure 8.9(a) shows the four possible rotor positions. The excitation sequence for stepping in this dual mode is as follows:

Circuits	Position	
A+ B- and C+ D-	1'	
A- B+ and C+ D-	2'	(excitation for Figure 8.9a)
A- B+ and C- D+	3'	
A+ B- and C- D+	4'	

**Figure 8.9**

Additional operating modes for stepper motors.



Having two circuits on at the same time produces considerably more torque than the single-excitation mode; however, more current is consumed, and the controller is more complex. Still, because it produces the greatest power-to-weight ratio, this bipolar, dual-excitation mode is quite common.

Both methods produce **four-step drives**, that is, four steps per cycle. By alternating the single- and dual-excitation modes, the motor can be directed to take **half-steps**, as shown in Figure 8.9(b). Positions 1, 2, 3, and 4 are from the single-excitation mode, and positions 1', 2', 3', and 4' are from the dual-excitation mode. When driven this way, the motor takes eight steps per revolution and is called an **eight-step drive**. This is desirable for some applications because it allows the motor to have twice the position resolution. Even smaller steps are possible with a process called *microstepping*, which is discussed later in the chapter.

The motor described thus far in this section steps  $90^\circ$  in the four-step mode (and  $45^\circ$  in the eight-step mode). PM stepper motors commonly have a smaller step, as low as  $30^\circ$ . This is done by increasing the poles in the rotor. Figure 8.11 shows a  $30^\circ$  stepper motor; the rotor has six rotor poles. Assume that field poles AB have been energized, pulling the rotor into the position shown. Next, circuit CD is energized. Rotor poles *yz* will be attracted to poles CD and will have to rotate only  $30^\circ$  to come into alignment.

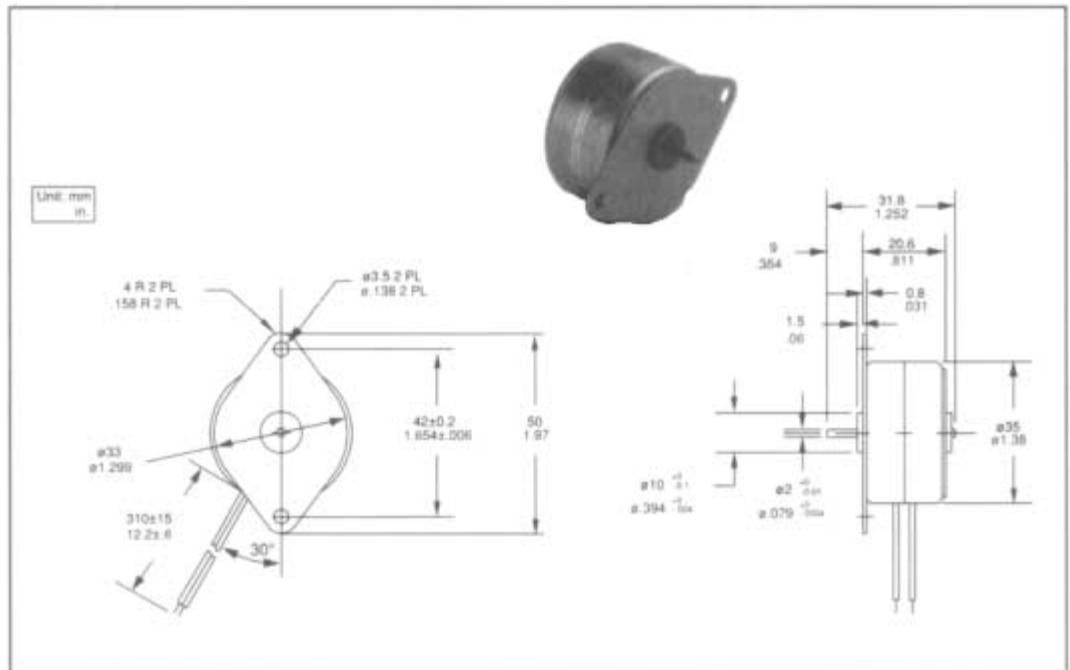
#### Four-Phase (Unipolar) Stepper Motors

The **four-phase** (unipolar) is the most common type of stepper motor (Figure 8.12). The term *four-phase* is used because the motor has four field coils that can be energized independently, and the term **unipolar** is applied because the current always travels in

**Figure 8.10**

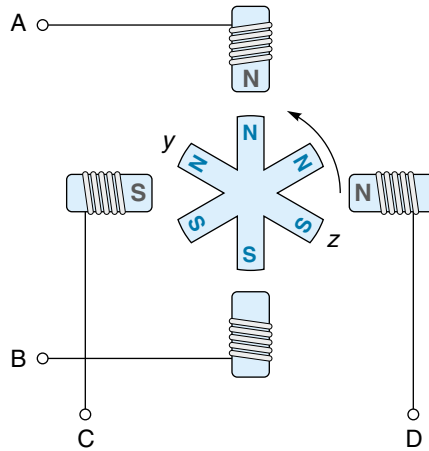
PM stepper motors.  
(Courtesy of  
Kollmorgen  
Motion  
Technologies  
Group)

PF35 Series	Models					
	PF35-48			PF35-24		
Excitation Mode	2-2					
Step Angle (°)	7.5			15		
Step Angle Tolerance (%)	±5					
Steps per Revolution	48			24		
Rating	Continuous			Intermittent	Continuous	
Letter Designator	C	D	Q	C	C	D
Winding Type	Unipolar	Unipolar	Bipolar	Unipolar	Unipolar	Unipolar
DC Operating Voltage (V)	12	5	5	24	12	5
Operating Current (mA/ø)	133	313	310	266	133	313
Winding Resistance (Ω/ø)	90	16	17	90	90	16
Winding Inductance (mH/ø)	48	8.9	12	48	48	8.9
Holding Torque (oz-in)	2.78		3.25	3.88	2.08	
Rotor Inertia (oz-in <sup>2</sup> )	24.1×10 <sup>-3</sup>					
Starting Pulse Rate, Max. (pps)	500		400	680	310	
Slewing Pulse Rate, Max. (pps)	530		500	770	410	
Ambient Temp Range, Operating (°C)	-10+50					
Temperature Rise (°C)	55			-	55	
Weight (oz)	2.8					



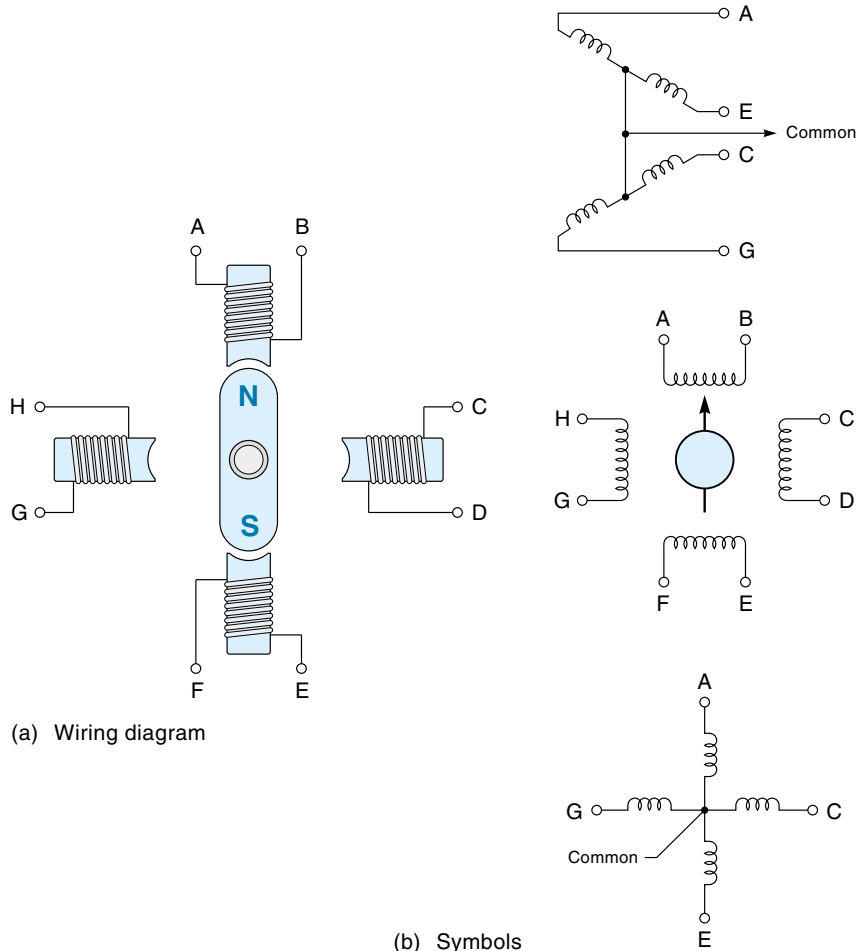
**Figure 8.11**

A 30 stepper motor with a six-pole rotor.



**Figure 8.12**

A four-phase (unipolar) stepper motor.



(a) Wiring diagram

(b) Symbols

the same direction through the coils. The simplest way to operate the four-phase stepper motor is to energize one phase at a time in sequence (known as *wave drive*). To rotate CW, the following sequence is used:

- A B
- C D (excitation for Figure 8.12a)
- E F
- G H

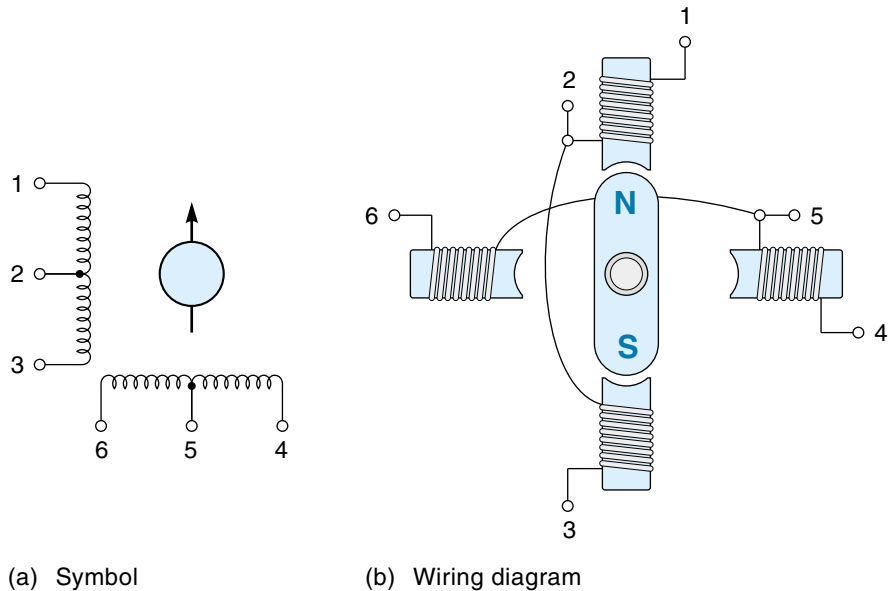
Compared with the two-phase bipolar motor, the four-phase motor has the advantage of simplicity. The control circuit of the four-phase motor simply switches the poles on and off in sequence; it does not have to reverse the polarity of the field coils. (However, the two-phase motor produces more torque because it is pushing and pulling at the same time.)

The torque of a four-phase stepper motor can be increased if two adjacent coils are energized at the same time, causing the rotor to align itself between the field poles [similar to that shown in Figure 8.9(a)]. Although twice the input energy is required, the motor torque is increased by about 40%, and the response rate is increased. By alternating single- and dual-excitation modes, the motor steps in half-steps, as shown in Figure 8.9(b).

Constructing motors so they can be used in either a two- or four-phase mode is common practice. This is done by bringing out two additional wires (from the two-phase motor) that are internally connected to points between the opposing field coils. Figure 8.13(a) shows the symbol for this type of motor, and Figure 8.13(b) shows the

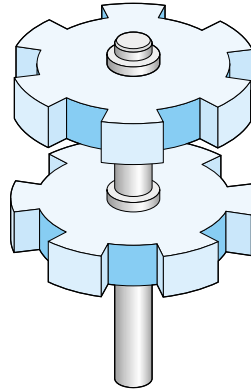
**Figure 8.13**

A four-phase stepper with center tap windings.



**Figure 8.14**

Stacked-rotor design allows smaller steps.



interior motor wiring. When such a motor is used in the two-phase mode, the center taps (terminals 2 and 5) are not used. When the motor is operated in the four-phase mode, the center taps become a *common return*, and power is applied to terminals 1, 4, 3, and 6 as required.

#### *Available PM Stepper Motors*

Almost all PM stepper motors available today have smaller step sizes than the simplified motors discussed so far. These motors are made by stacking two multipolled rotors (offset by one-half step), as illustrated in Figure 8.14. Typical step sizes for four-phase PM stepper motors are  $30^\circ$ ,  $15^\circ$ , and  $7.5^\circ$ . Figure 8.10 is a specification sheet for a family of PM stepper motors. For example, the PF35-48C (continuous duty) motor has 48 steps/revolution (which is  $7.5^\circ$ ), requires 133 mA at 12 V, and has a holding torque of 2.78 in. · oz.

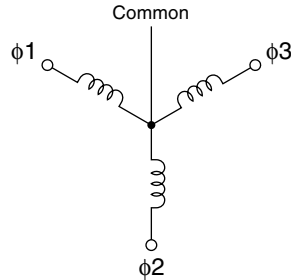
## 8.2 VARIABLE-RELUCTANCE STEPPER MOTORS

The **variable-reluctance (VR) stepper motor** does not use a magnet for the rotor; instead, it uses a toothed iron wheel [see Figure 8.15(b)]. The advantage of not requiring the rotor to be magnetized is that it can be made in any shape. Being iron, each rotor tooth is attracted to the closest energized field pole in the stator, but not with the same force as in the PM motor. This gives the VR motor less torque than the PM motor.

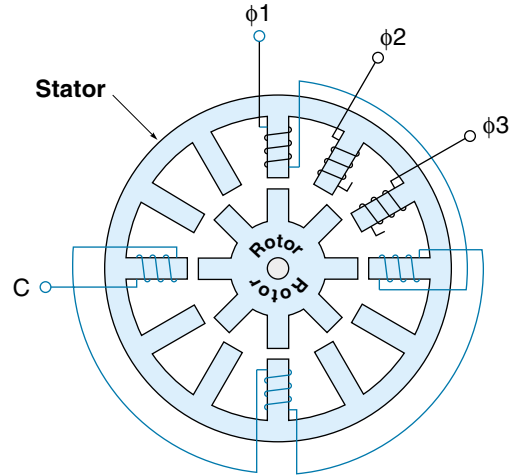
A VR motor usually has three or four phases. Figure 8.15(a) shows a typical **three-phase stepper motor**. The stator has three field pole circuits:  $\emptyset 1$ ,  $\emptyset 2$ , and  $\emptyset 3$ . Figure 8.15(b) shows that the actual motor has 12 field poles, where each circuit energizes four windings; you can see this by closely observing the  $\emptyset 1$  wire in Figure 8.15(b). Notice that the rotor has only 8 teeth even though there are 12 teeth in the stator. Therefore, the rotor teeth can never line up “one for one” with the stator teeth, a fact that plays an important part in the motor’s operation.

**Figure 8.15**

A three-phase VR stepper motor (15 step). (Wires for  $\Delta 2$  and  $\Delta 3$  left out for clarity.)



(a) Symbol



(wires for  $\phi 2$  &  $\phi 3$  left out for clarity)

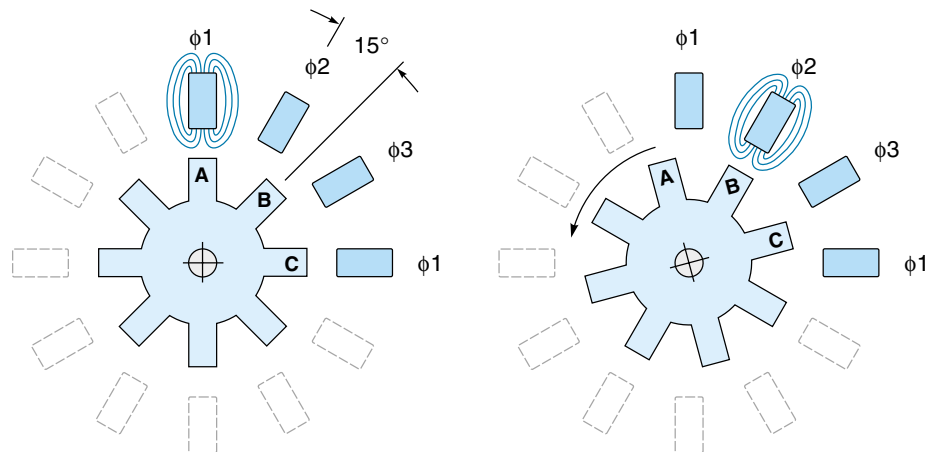
(b) Construction

Figure 8.16 illustrates the operation of the VR stepper motor. When circuit  $\phi 1$  is energized, the rotor will move to the position shown in Figure 8.16(a)—that is, a rotor tooth (A) is lined up with the  $\phi 1$  field pole. Next, circuit  $\phi 2$  is energized. Rotor tooth B, being the closest, is drawn toward it [Figure 8.16(b)]. Notice that the rotor has to move only  $15^\circ$  for this alignment. If circuit  $\phi 3$  is energized next, the rotor would continue CCW another  $15^\circ$  by pulling tooth C into alignment.

The step angle of the VR motor is the difference between the rotor and stator angles. For the motor of Figure 8.16, the angle between the field poles is  $30^\circ$ , and the

**Figure 8.16**

A  $15^\circ$  three-phase VR stepper motor (only four field poles shown for clarity).



(a) Pole  $\phi 1$  energized

(b) Pole  $\phi 2$  energized

angle between the rotor poles is  $45^\circ$ . Therefore, the step is  $15^\circ$  ( $45^\circ - 30^\circ = 15^\circ$ ). By using this design, the VR stepper motor can achieve very small steps (less than  $1^\circ$ ). Small step size is often considered to be an advantage because it allows for more precise positioning.

The VR stepper motor has a number of functional differences when compared with the PM type. Because the rotor is not magnetized, it is weaker (less torque) than a similar-sized PM stepper motor. Also, it has no detent torque when the power is off, which can be an advantage or disadvantage depending on the application. Finally, because of the small step size and reduced detent torque, the VR stepper motor has more of a tendency to overshoot and skip a step. This is a serious matter if the motor is being operated open-loop, where position is maintained by keeping track of the number of steps taken. To solve the problem, some sort of damping may be required. This can be done mechanically by adding friction or electrically by providing a slight braking torque with adjacent field poles.

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## 8.3 HYBRID STEPPER MOTORS

The **hybrid stepper motor** combines the features of the PM and VR stepper motors and is the type in most common use today. The rotor is toothed, which allows for very small step angles (typically  $1.8^\circ$ ), and it has a permanent magnet providing a small detent torque even when the power is off.

Recall that the step size of a PM motor is limited by the difficulty in making a multipole magnetized rotor. There is simply a limit to the number of different magnetizations that can be imposed on a single iron rotor. The VR stepper motor gets around this by substituting iron teeth (of which there can be many) for magnetized poles on the rotor. This approach allows for a small step angle, but it sacrifices the strength and detent torque qualities of the PM motor. The hybrid motor can effectively magnetize a multitoothed rotor and thus has the desirable properties of both the PM and VR motors.

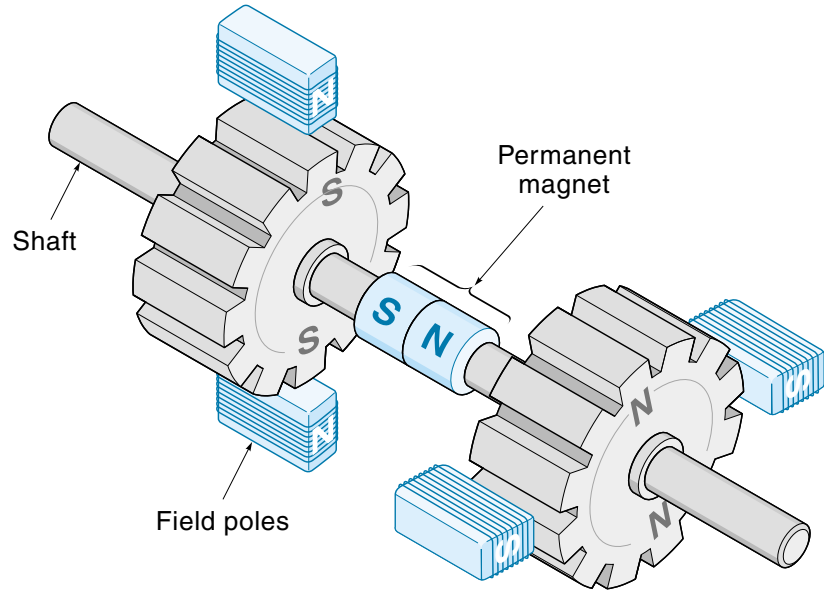
Figure 8.17 illustrates the internal workings of the hybrid motor, which is considerably more complicated than the simple PM motor. The rotor consists of two toothed wheels with a magnet in between—one wheel being completely north in magnetization and the other being completely south. For each step, two opposing teeth on the north wheel are attracted to two south field poles, and two opposing teeth on the south wheel are attracted to two north field poles. The internal wiring is more complicated than it is for the PM or VR motors, but to the outside world this motor is just as simple to control.

The theory of operation of the hybrid motor is similar to the VR motor in that the rotor and stator have a different number of teeth, and for each step, the closest energized teeth are pulled into alignment. However, the principles of magnetics require that, at any one time, half the poles be north and the other half be south. To maintain the



**Figure 8.17**

Internal construction of the hybrid stepper motor (only two poles per stator shown for clarity).



magnetic balance, each pole must be able to switch polarity so that it can present the correct pole at the correct time. This is accomplished in one of two ways: For a bipolar motor, the applied voltage must be reversed by the driver circuit (similar to the two-phase PM motor). On the other hand, a unipolar motor has two separate windings of opposite direction on each field pole (called a *bifilar winding*), so each pole can be a north or a south. Therefore, the unipolar hybrid motor does not require a polarity-reversing driver circuit.

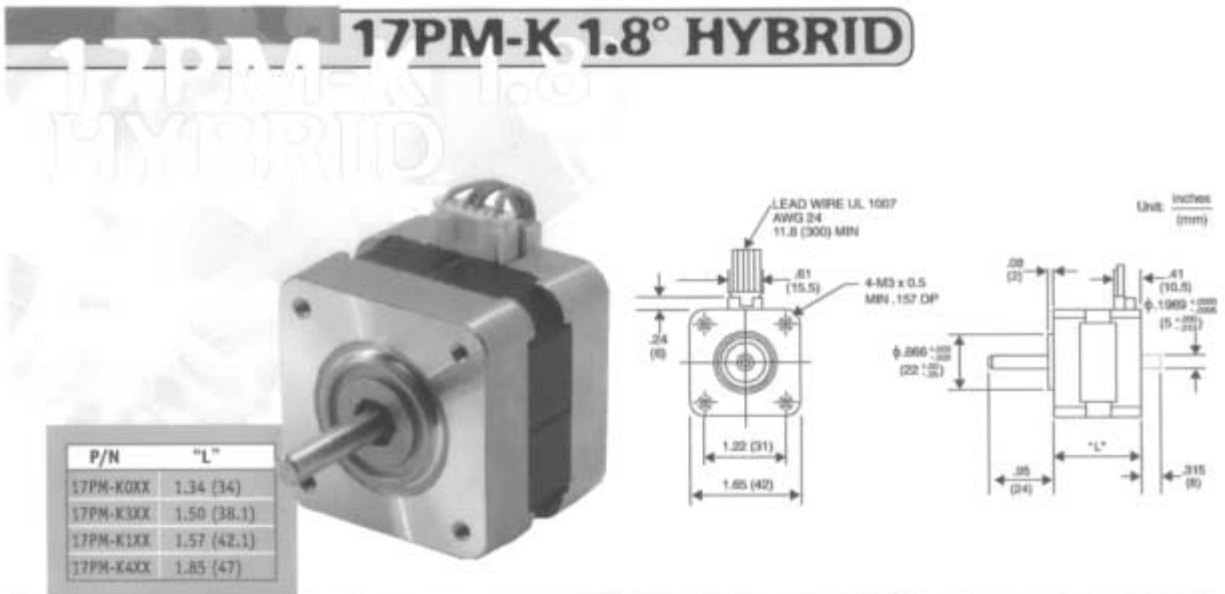
Figure 8.18 shows a specification sheet for a family of  $1.8^\circ$  hybrid stepping motors. For example, the 17PM-K016V draws 0.4A at 8.8 V and has a holding torque of 1500 gm · cm (20.8 in. · oz) and a detent torque of 80 gm · cm (1.1 in.-oz). (Note: There are 72 gm · cm per in. · oz.)

## 8.4 STEPPER MOTOR CONTROL CIRCUITS

Figure 8.19 shows the block diagram for a stepper motor driving circuit. The *controller* decides on the number and direction of steps to be taken (based on the application). The *pulse sequence generator* translates the controller's requests into specific stepper motor coil voltages. The *driver amplifiers* boost the power of the coil drive signals. It should be clear that the stepper motor is particularly well suited for digital control; it requires no digital-to-analog conversion, and because the field poles are either on or off, efficient class C driver amplifiers can be used.

**Figure 8.18**

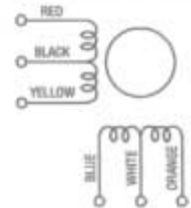
Hybrid stepper motors.  
(Courtesy of NMB Corp.)



**GENERAL SPECIFICATIONS**

Step Angle ..... 1.8°  
 Step Angle Accuracy ..... +/- 5%  
 Temperature Rise ..... 80° C Max.  
 Ambient Temperature Range ..... -20° to +50° C  
 Insulation Resistance ..... 100MΩ Min., 500 VDC  
 Dielectric Strength ..... 500 VAC for 1 min.  
 Radial Play ..... 0.02 mm Max. (450 g-load)  
 End Play ..... 0.08 mm Max. (450 g-load)  
 Switching Sequence ..... See page 31

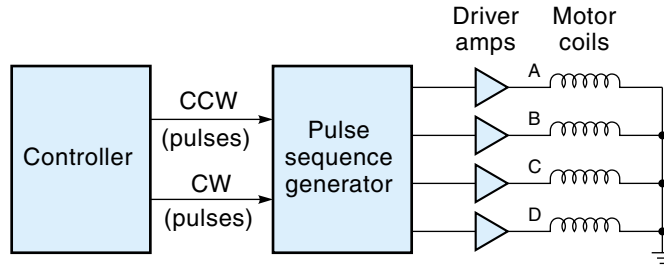
**WINDING DIAGRAM**



**MODEL SPECIFICATIONS**

Model Number	Rated Voltage V	Rated Current/ Phase A	Winding Resistance/ Phase Ω	Holding Torque g-cm	Inductance mH	Rotor Inertia g-cm <sup>2</sup>	Detent Torque g-cm	Weight g
17PM-K016V	8.80	0.40	22.00	1,500	19.5	34.0	80	200
17PM-K017V	4.40	0.80	5.50	1,500	5.7	34.0	80	200
17PM-K018V	3.00	1.20	2.50	1,500	2.8	34.0	80	200
17PM-K316V	9.60	0.40	24.00	1,700	25.8	45.0	100	250
17PM-K301V	4.80	0.80	6.00	1,700	7.1	45.0	100	250
17PM-K303V	3.20	1.20	2.70	1,700	3.3	45.0	100	250
17PM-K111V	10.00	0.40	25.00	2,200	33.4	56.0	120	300
17PM-K101V	5.00	0.80	6.20	2,200	8.6	56.0	120	300
17PM-K103V	3.60	1.20	3.00	2,200	4.4	56.0	120	300
17PM-K402V	6.00	0.80	7.50	3,400	7.0	75.0	200	350

**Figure 8.19**  
Block diagram of  
stepper motor  
control circuit.

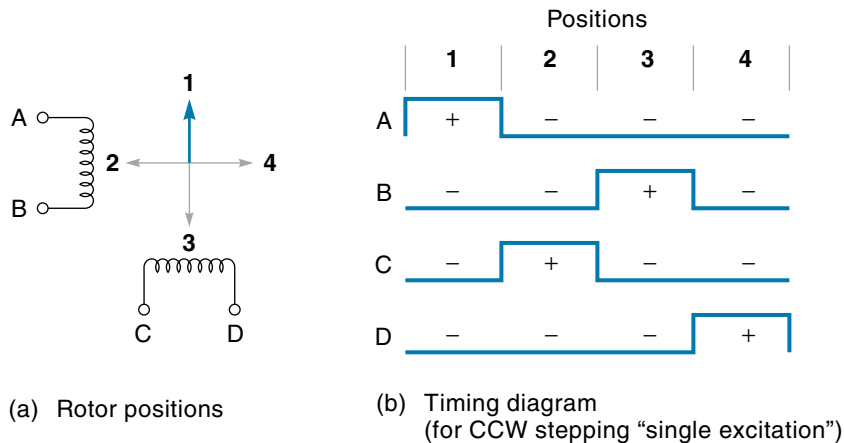


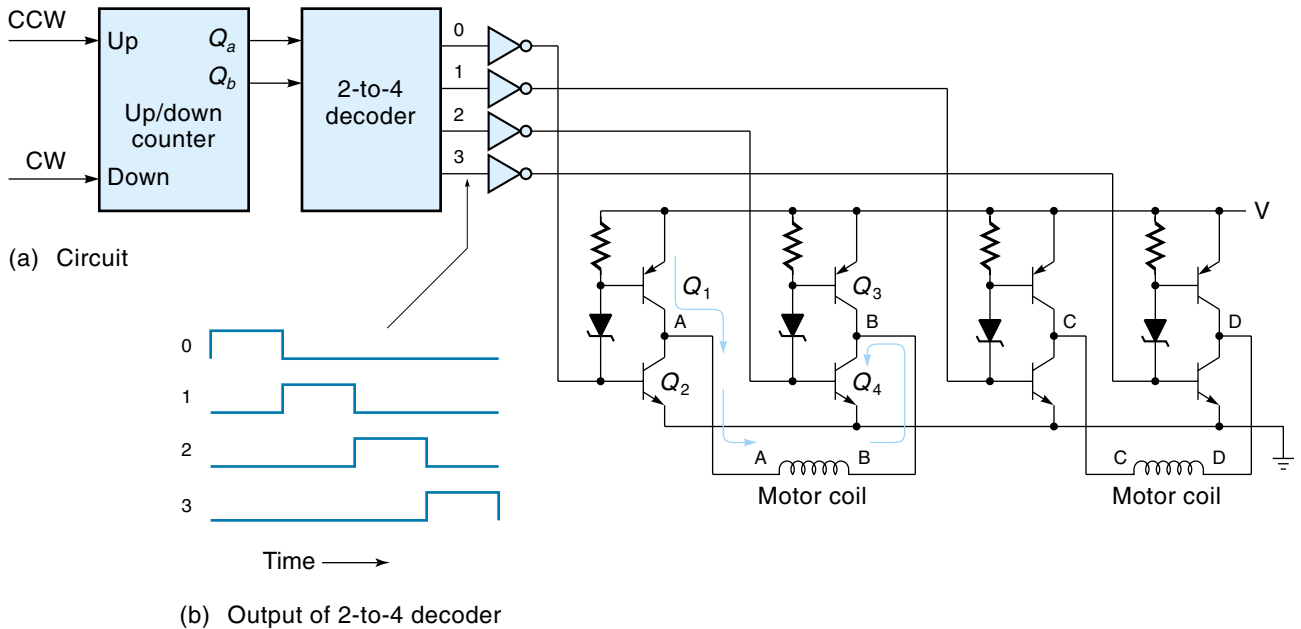
### Controlling the Two-Phase Stepper Motor

Controlling the two-phase bipolar stepper motor requires polarity reversals, making it more complicated than four-phase motor controllers. Figure 8.20 shows a two-phase stepper motor. The two circuits are designated AB and CD. The timing diagram shows the required waveforms for A, B, C, and D (CCW rotation). Looking down the position 1 column in Figure 8.20(b), we see A is positive and B is negative, so current will flow from A to B in circuit AB. Meanwhile C and D are both negative, effectively turning off circuit CD. For position 2 in the timing diagram, C is positive, and D is negative; causing current to flow from C to D in circuit CD while coil AB is completely off, and so on, for positions 3 and 4.

A digital circuit such as the one shown in Figure 8.21(a) can be used to generate the timing waveforms. The up/down counter is a 2-bit counter that increments for each pulse received on its Up input and decrements for each pulse received on the Down input.  $Q_a$  and  $Q_b$  of the up/down counter are decoded in a 2-to-4 decoder. As the counter is always in one of four states (00, 01, 10, 11), one (and only one) of the four decoder outputs is “high” at any particular time. Figure 8.21(b) shows the output of the decoder when the counter counts up (a result of CCW pulses from the controller).

**Figure 8.20**  
Two-phase (bipolar)  
stepper motor  
operation.



**Figure 8.21**

Complete interface circuit for a two-phase (bipolar) stepper motor.

The next task is to connect the timing signals from the decoder in such a way as to drive the motor coils. This can be accomplished with the power amplifier circuit shown on the right side of Figure 8.21(a). Notice there are four complementary-symmetry drivers, one for each end of each motor coil. When  $Q_1$  and  $Q_4$  are on, the current can flow through the motor in the direction shown (left to right). On the other hand, when  $Q_3$  and  $Q_2$  are on, the polarity is reversed, and the current flows the opposite direction through the motor (right to left). Finally, if  $Q_1$  and  $Q_3$  are off, no current flows through the motor coil.

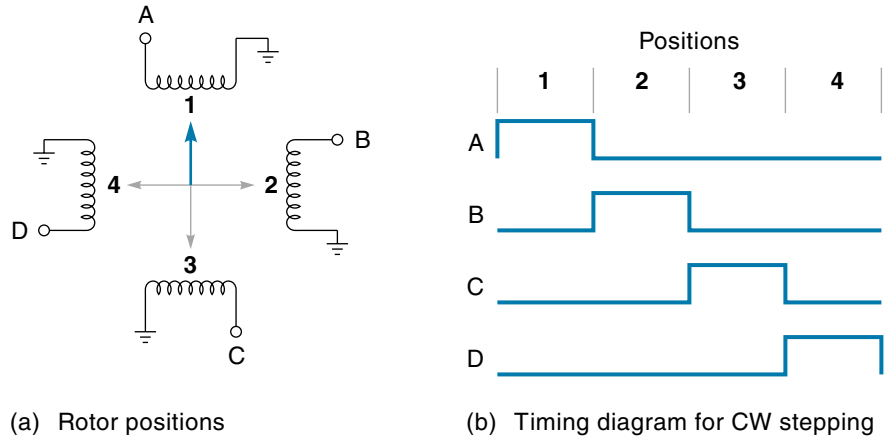
The four outputs of the decoder (which must be inverted in this case) control the four complementary-symmetry transistor circuits. The resistor and zener diode in each circuit cause the upper transistor to be on when the lower transistor is off, and vice versa. Trace through the circuit for each step of the decoder, and you will see that the timing diagram of Figure 8.20(b) is reproduced. This arrangement will provide for the motor to step CCW when the counter counts up. When the counter counts down, the sequence will be backward, and the motor will step CW.

### Controlling the Four-Phase Stepper Motor

The electronics needed to drive the four-phase stepper motor is simpler than for a two-phase motor because polarity reversals are not required. Figure 8.22 identifies the coils in a four-phase motor and shows the timing diagram for simple single-excitation CW

**Figure 8.22**

Four-phase (unipolar) stepper motor operation.

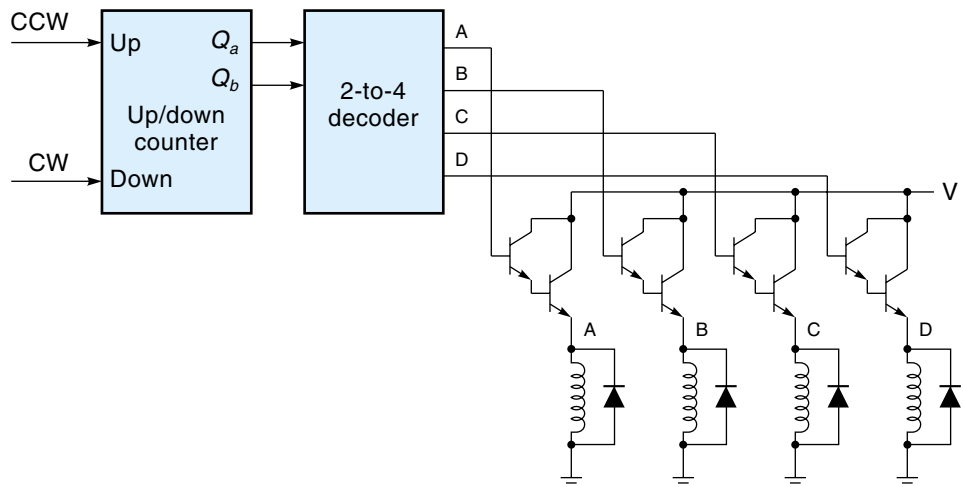


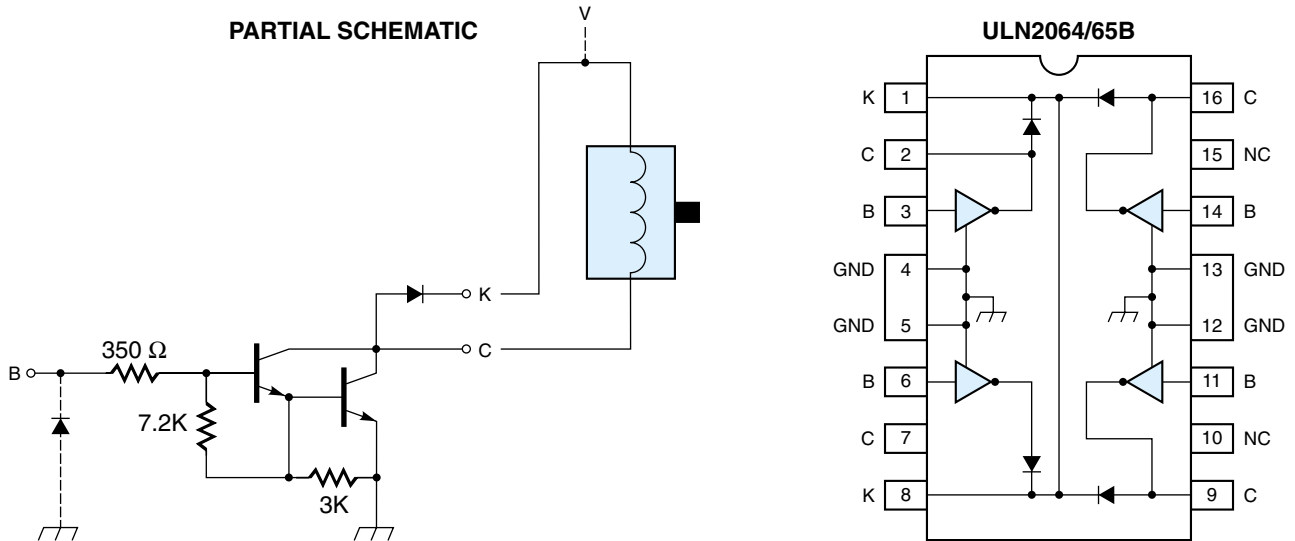
stepping. The timing is very straightforward and can easily be generated with the counter and decoder circuit shown in Figure 8.23.

The outputs of the decoder (Figure 8.23) are connected to four Darlington driver transistors. As timing signals A, B, C, and D go high in sequence, the corresponding transistors are turned on, energizing coils in the motor. The necessary four Darlings are available on a single IC such as the Allegro ULN-2064/65B (Figure 8.24). These amplifiers can supply up to 1.5 A and can be driven directly from TTL 5-V logic. (Note the diode to allow an escape path for current in the coil when the transistor is turned off.)

**Figure 8.23**

Complete interface for a four-phase stepper motor (simplified for clarity).





**Figure 8.24**  
The Allegro ULN-2064B with four Darlington 1.5-A switches. (Courtesy of Allegro MicroSystems)

ICs designed specifically to drive stepper motors contain both the timing logic and power drivers in one package. One example is the Allegro UCN-5804B (Figure 8.25). The basic inputs are the step input (pin 11) and direction (pin 14). The motor will advance one step for each pulse applied to the step input pin, and the logic level on the direction pin determines if rotation will be CW or CCW. Notice that the output transistors are in the common emitter configuration (called *open collector*). The motor coils should be connected between the output pins and the supply voltage (as shown). When an output transistor turns on, it completes the circuit by providing a path to ground for the motor coil current. The three operating modes of the UCN-5804B are given in tabular form in Figure 8.25(b), and will be explained using Figure 8.26:

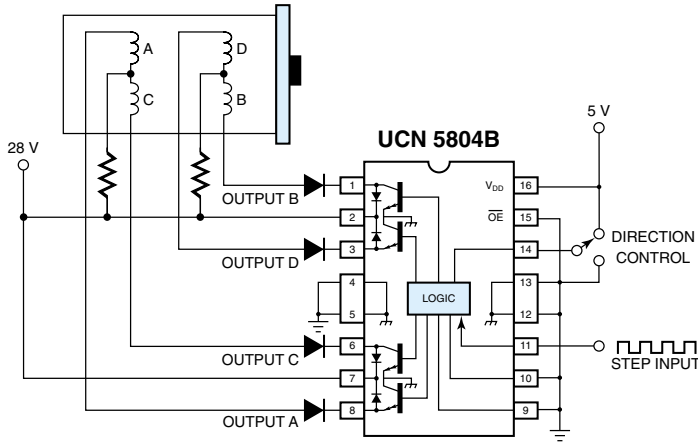
1. Figure 8.26(a) shows how the motor responds to the wave-drive sequence, a single-excitation mode where the coils A, B, C, and D are energized one at a time in sequence.
2. Figure 8.26(b) shows how the motor responds to the two-phase drive sequence, a dual-excitation mode where two adjacent phases are energized at the same time to give more torque.
3. Figure 8.26(c) shows how the motor responds to the half-step drive sequence, where the operation alternates between single- and dual-excitation modes, yielding eight half-steps per cycle.

## Microstepping

**Microstepping**, a technique that allows a stepper motor to take fractional steps, works by having two adjacent field poles energized at the same time, similar to half-steps

**Figure 8.25**

A unipolar stepper motor translator/driver (Allegro UCN-5804B).  
(Courtesy of Allegro MicroSystems)



**WAVE-DRIVE SEQUENCE**

Half Step = L, One Phase = H				
Step	A	B	C	D
POR	ON	OFF	OFF	OFF
1	ON	OFF	OFF	OFF
2	OFF	ON	OFF	OFF
3	OFF	OFF	ON	OFF
4	OFF	OFF	OFF	ON

**TWO-PHASE DRIVE SEQUENCE**

Half Step = L, One Phase = L				
Step	A	B	C	D
POR	ON	OFF	OFF	ON
1	ON	OFF	OFF	ON
2	ON	ON	OFF	OFF
3	OFF	ON	ON	OFF
4	OFF	OFF	ON	ON

**HALF-STEP DRIVE SEQUENCE**

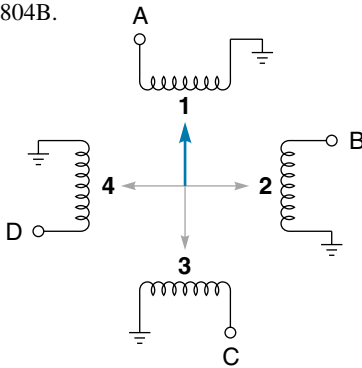
Half Step = H, One Phase = L				
Step	A	B	C	D
POR	ON	OFF	OFF	OFF
1	ON	OFF	OFF	ON
2	ON	ON	OFF	OFF
3	OFF	ON	OFF	OFF
4	OFF	ON	ON	OFF
5	OFF	OFF	ON	OFF
6	OFF	OFF	ON	ON
7	OFF	OFF	OFF	ON
8	ON	OFF	OFF	ON

(a) Driver circuit

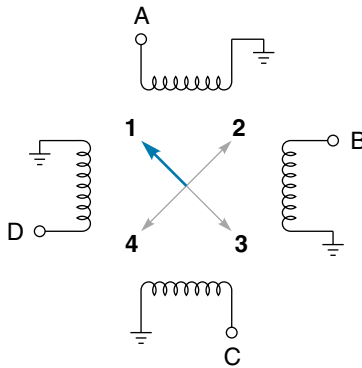
(b) Modes of operation

**Figure 8.26**

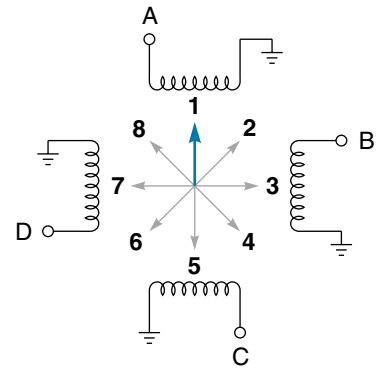
Three modes of operation for the Allegro UCN-5804B.



(a) Wave-drive

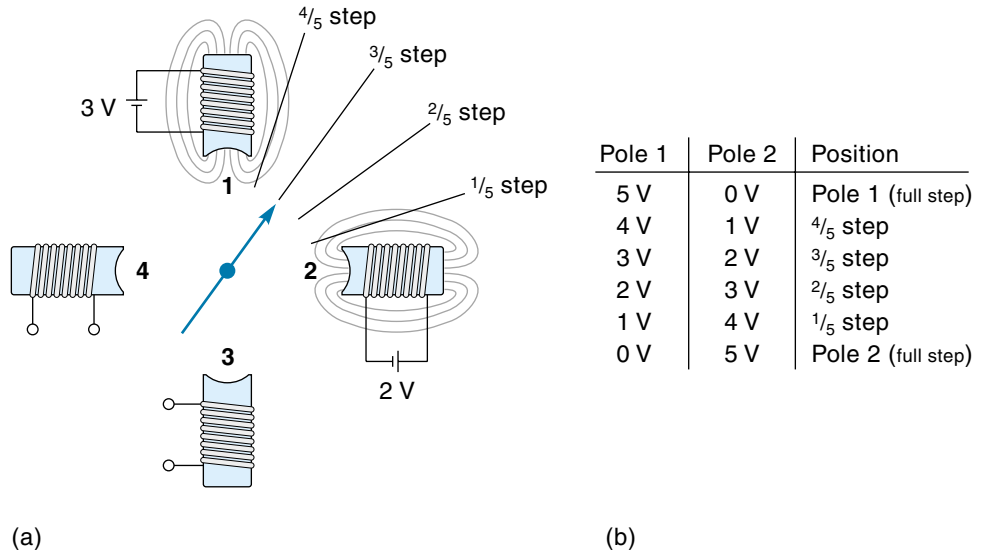


(b) Two-phase



(c) Half-step

**Figure 8.27**  
Microstepping.



described earlier. In microstepping the adjacent poles are driven with different voltage levels, as demonstrated in Figure 8.27(a). In this case, pole 1 is supplied with 3 V and pole 2 with 2 V, which causes the rotor to be aligned as shown—that is, three-fifths of the way to pole 1. Figure 8.27(b) shows the voltages (for poles 1 and 2) to get five microsteps between each “regular” step. The different voltages could be synthesized with pulse-width modulation (PWM). The most commonly used microstep increments are  $1/5$ ,  $1/10$ ,  $1/16$ ,  $1/32$ ,  $1/125$ , and  $1/250$  of a full step. Another benefit of microstepping (for delicate systems) is that it reduces the vibrational “shock” of taking a full step—that is, taking multiple microsteps creates a more “fluid” motion.

Two other points on microstepping: It does not require a special stepper motor, only special control circuitry, and the actual position of the rotor (in a microstepping system) is very dependent on the load torque.

### Improving Torque at Higher Stepping Rates

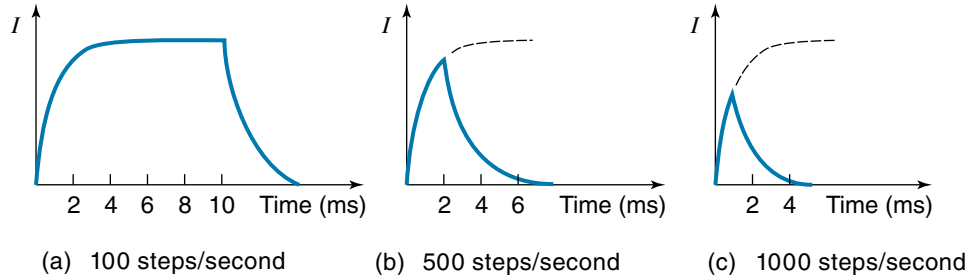
It is important that the stepper motor develop enough torque with each step to drive the load. If it doesn't, the motor will stall (not step). When steps are missed, the controller no longer knows the exact position of the load, which may render the system useless.

At higher stepping rates, two problems occur. First, if the load is accelerating, extra torque is needed to overcome inertia; second, the available motor torque actually diminishes at higher speeds. Recall that motor torque is directly proportional to motor current and that the average current decreases as the stepping rate increases. This is illustrated in Figure 8.28, which shows the motor current for three stepping rates. The problem is that the rate of change of current is limited by the circuit-time constant  $\tau$ .



**Figure 8.28**

Coil current as a function of stepping speed.



$$\tau = \frac{L}{R} \tag{8.1}$$

where

$\tau$  = time constant

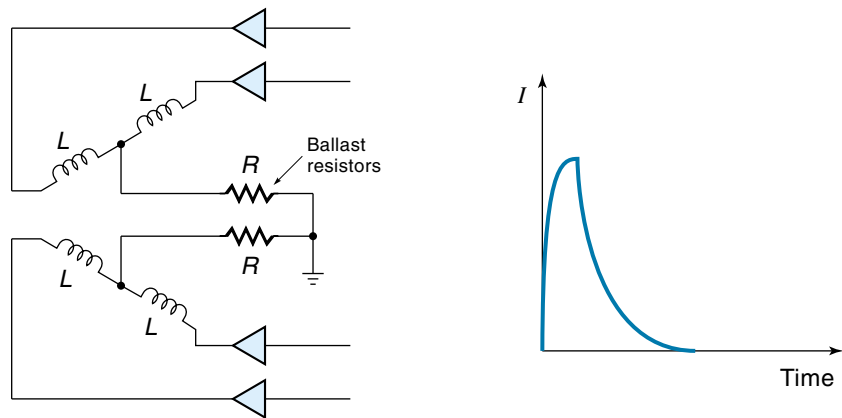
$L$  = motor inductance

$R$  = motor coil resistance

You can see that as the stepping rate increases the current cannot build up in the field coils to as great a value. If we could reduce the value of the motor-time constant, the current could build up faster. One way to do this would be to increase the value of  $R$  in Equation 8.1. This can easily be done by adding external resistors ( $R$ ) in series with the motor coils as shown in Figure 8.29(a). Such resistors are called **ballast resistors**, and their purpose is to improve the torque output at higher stepping rates (it also limits the current, which may be important in some cases). Stepper motor driver circuits that use ballast resistors are called **L/R drives**. Figure 8.29(b) shows the motor current with the ballast resistor added (for a rate of 1000 steps/second). Compare this with the last graph in Figure 8.28 to see the improvement.

**Figure 8.29**

Effect of adding an external ballast resistor.

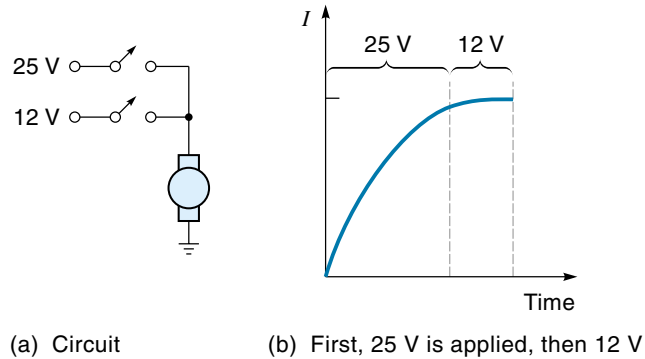


(a) Motor coils and ballast resistors

(b) Coil current at 1000 steps/second with ballast resistor

**Figure 8.30**

Principle of a bilevel drive.



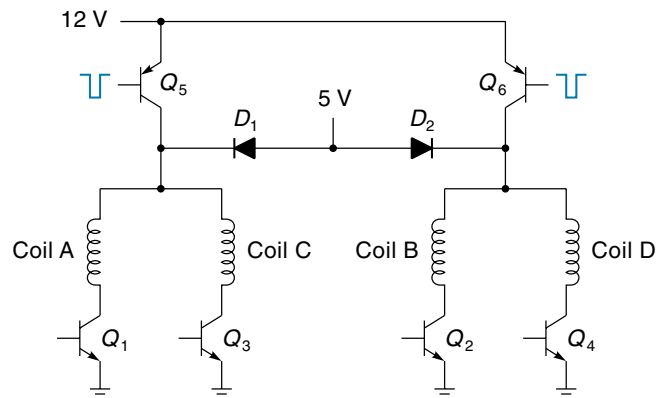
Another way to improve motor torque at higher stepping rates is to use **bilevel drive**. In this approach, a high voltage is momentarily applied to the motor at the beginning of the step to force a fast in-rush of current. Then a lower voltage level is switched on to maintain that current. Figure 8.30(a) shows a simplified circuit to provide bilevel drive. The 25-V circuit is switched on, and the current rises rapidly [Figure 8.30(b)]. When the desired current level is reached, the 25-V circuit is switched off, and the 12-V circuit is switched on, which keeps the current at the desired level for the rest of the step time.

Figure 8.31 shows a bilevel-drive interface circuit. In this case, the higher voltage is 12 V, and the lower voltage is 5 V. The 12-V is switched through either  $Q_5$  or  $Q_6$  in response to a pulse from a timing circuit (not shown). The 5-V is applied through  $D_1$  and  $D_2$ . These diodes keep the 12-V pulses from backing up into the 5-V power supply. The bilevel drive is more complex but allows the stepper motor to have more torque at higher stepping rates.

Another approach for providing higher torque at faster stepping rates is the **constant current chopper drive**. Using PWM techniques, this driver circuit can deliver almost the same current to the motor at all speeds. A chopper-drive waveform is shown in Figure 8.32 and works in the following manner: A relatively high voltage is switched to the motor coil, and the current is monitored. When the current reaches a specified level, the voltage is cut off. After a short time, the voltage is reapplied, and the current

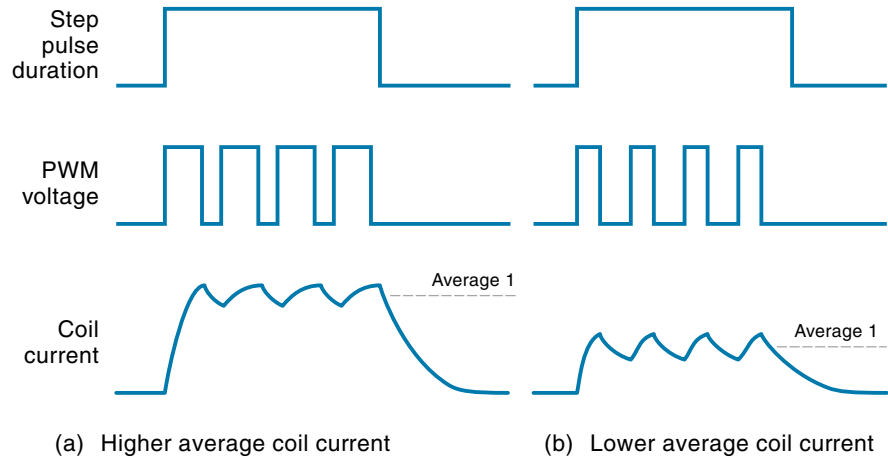
**Figure 8.31**

A bilevel-drive circuit.



**Figure 8.32**

A PWM (chopper drive) used to regulate stepper motor coil current.



again increases, only to be cut off, and so on. Thus, in the same way that a thermostat can maintain a constant temperature by switching the furnace on and off, the chopper drive maintains a constant average current (within each drive pulse) by rapidly switching the voltage on and off. In summary, the chopper drive is another technique for providing good torque at high stepping rates. Stepper motor driver ICs are available, such as the Allegro A2919SB, with built-in PWM constant current capability.

## 8.5 STEPPER MOTOR APPLICATION: POSITIONING A DISK DRIVE HEAD

Example 8.3 illustrates many of the principles presented in this chapter and extends the discussion to show how software can control a stepper motor.

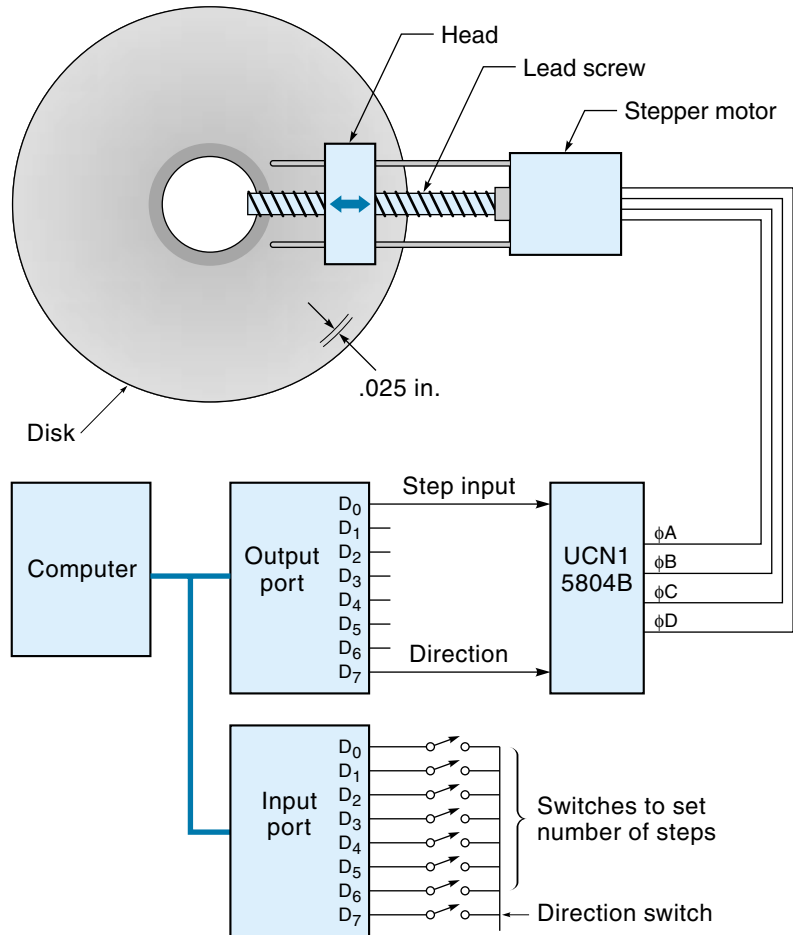
### EXAMPLE 8.3

A 30° four-phase stepper motor drives the read/write head on a floppy disk drive (Figure 8.33). The in-and-out linear motion is achieved with a leadscrew connected directly to the motor shaft. Each magnetic track on the disk is 0.025 in. apart (40 tracks/in.). The leadscrew has 20 threads/in.

The motor is driven by the UCN-5804B stepper motor interface IC. This IC requires only two inputs: step input and direction. A computer will supply these signals in response to toggle switch settings. A front panel contains eight toggle switches; seven are used to input (in binary) the number of tracks to move, and the eighth switch specifies direction—in or out. Write a program in BASIC that will cause the motor to step the number of tracks and direction

**Figure 8.33**

The hardware setup for the stepper motor (Example 8.3).



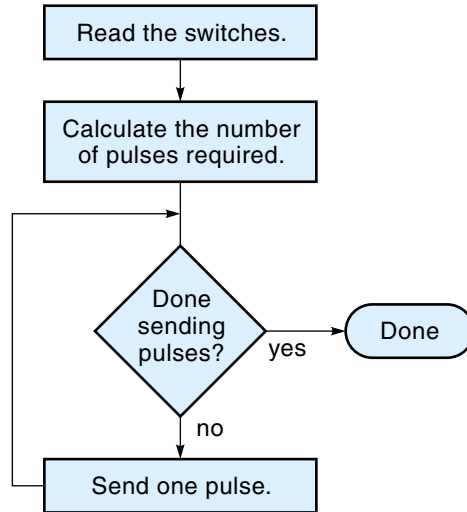
specified by the switch settings. Assume your computer hardware has an 8-bit input port and an 8-bit output port.

### SOLUTION

First we need to find the number of steps required to advance one track on the disk. If the leadscrew has 20 threads per inch, then rotating it one revolution ( $360^\circ$ ) will advance it 1 thread, which is  $\frac{1}{20}$  in. The following equation was set up by multiplying all the component transfer functions, including conversion factors as necessary (and oriented so that, if possible, the units would cancel):

**Figure 8.34**

Flowchart for a program to drive a stepper motor.



$$\frac{0.025 \text{ in.}}{\text{track}} \times \frac{20 \text{ threads}}{\text{in.}} \times \frac{360^\circ}{\text{thread}} \times \frac{1 \text{ step}}{30^\circ} = \frac{6 \text{ steps}}{\text{track}}$$

Thus, the stepper motor must take six steps to advance one track on the disk.

The program must first read the switches, then calculate the required number of steps, and, finally, output the step command pulses, one by one, to the UCN-5804B. The direction bit must also be read and passed along to the UCN-5804B. Figure 8.34 shows a simplified flowchart of the program.

The next step is to translate the flowchart into a BASIC program. The complete program, along with line-by-line explanations, is given in Table 8.1. As shown in Figure 8.33, the input port receives the 8 bits from the switches. For output, only 2 of the 8 bits are used: the least significant bit (LSB) ( $D_0$ ) for the step input pulse and the most significant bit (MSB) ( $D_7$ ) for the direction command. The step input pulse is created by making the output ( $D_0$ ) go “high” for a period of time and then bringing it “low” for a period of time. The direction bit, brought in as the MSB from the switches, is simply passed on through by the program and sent out as the MSB (actually it is stripped off the input data and added back later to the output word). The parallel I/O port addresses would, of course, depend on the system being used; in this case, the input switch port is 209 (decimal), and the output port to the motor has an address of 208 (decimal).

TABLE 8.1

## BASIC Program to Control a Disk Drive Head

	Instruction	Explanation
10	SW = INP(209)	Input 8 bits of switch data.
20	IF SW < 128 THEN DIR = 0 ELSE DIR = 128 SW = SW - 128 END IF	If MSB = 0, then the direction bit = 0.  Otherwise, set direction bit = 1 (10000000 = 128) and remove the MSB to leave the number of tracks.
30	PUL = SW*6	Calculate the number of pulses needed at 6 pulses per track.
40	REM***** SEND PULSES*****	
50	FOR I = 1 TO PUL	Prepare to send PUL (number of pulses).
60	HIGH = 1 + DIR	Make the LSB a 1 and add a direction bit.
70	OUT 208, 1	Send out a "high" for the step pulse.
80	FOR J=1 TO 100	This is the time-delay loop for the pulse width.
90	NEXT J	
100	LOW = 0 + DIR	Make the LSB a 0 and add a direction bit.
110	OUT 208, 0	Send out a "low" to define the end of the pulse.
120	FOR J = 1 TO 100	This is the time-delay loop for the pulse being "low."
130	NEXT J	
140	NEXT I	Go back and send the next pulse.
	END	

## SUMMARY

A stepper motor, a unique type of DC motor, rotates in fixed steps of a certain number of degrees: 30°, 15°, 1.8°, and so on. The rotor (the part that moves) is made of permanent magnets or iron and contains no coils and therefore has no brushes. Surrounding the rotor is the stator, which contains a series of field pole electromagnets. As the electromagnets are energized one after the other, the rotor is pulled around in a circle. Stepper motors used in position systems are usually operated open-loop, that is, without feedback sensors. The controller will step the motor so many times and expect the motor to be there.

The permanent magnet (PM) stepper motor uses permanent magnets in the rotor. This type of motor has a detent torque—a small magnetic tug that tends to keep it in the last position stepped to, even when the power is removed. PM motors have good torque capability but cannot take very small steps. The field coils of the PM motor can

be driven in the two-phase or four-phase mode. The two-phase mode requires polarity reversals. Four-phase operation does not require polarity reversals and the timing is more straightforward.

The variable reluctance (VR) stepper motor uses a toothed iron wheel for the rotor (instead of permanent magnets). This allows VR motors to take smaller steps, but they are weaker and have no detent torque.

The hybrid stepper motor combines the features of both the PM and VR stepper motors and is the type in most common use today. Hybrid motors can take small steps (typically  $1.8^\circ$ ) and have a detent torque. The internal construction of the hybrid motor is more complicated than the PM or VR motors, but the electrical operation is just as simple.

Stepper motors are driven from digital circuits that provide the desired number of stepping pulses (in the correct order) to the field coils. These pulses must usually be amplified with driver transistors (operating as switches) before being applied to the motor coils. ICs are available that can provide the proper sequencing and amplification in one chip.

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## GLOSSARY

**ballast resistor** A resistor placed in series with the motor coils to improve the torque at higher stepping rates; it works by reducing the motor-time constant.

**bilevel drive** A technique that uses two voltages to improve torque at higher stepping rates. A higher voltage is applied to the motor at the beginning of the step, and then a lower voltage is switched in.

**bipolar motor** A motor that requires polarity reversals for some of the steps; a two-phase motor is bipolar.

**constant current chopper drive** A drive circuit for stepper motors that uses PWM techniques to maintain a constant average current at all speeds.

**cumulative error** Error that accumulates; for example, a cumulative error of  $1^\circ$  per revolution means that the measurement error would be  $5^\circ$  after five revolutions.

**detent torque** A magnetic tug that keeps the rotor from turning even when the power is off; also called *residual torque*.

**dynamic torque** The motor torque available to rotate the load under normal conditions.

**eight-step drive** A two- or four-phase motor being driven in half-steps; the sequencing pattern has eight steps.

**four-phase stepper motor** A motor with four separate field circuits; this motor does not require polarity reversals to operate and hence is unipolar.

**four-step drive** The standard operating mode for two- or four-phase stepper motors taking full steps; the sequencing pattern has four states.

**half-steps** By alternating the standard mode with dual excitation mode, the angle of step will be half of what it normally is.

**holding torque** The motor torque available to keep the shaft from rotating when the motor is stopped but with the last field coil still energized.

**hybrid stepper motor** A motor that combines the features of the PM and VR stepper motors—that is, it can take small steps and has a detent torque.

**L/R drive** A stepper motor driver circuit that uses ballast resistors in series with the motor coils to increase torque at higher stepping rates.

**microstepping** A technique that allows a regular stepper motor to take fractional steps; it works by energizing two adjacent poles at different voltages and by balancing the rotor between.

**permanent-magnet (PM) stepper motor** A motor that uses one or more permanent magnets for the rotor; this motor has a detent torque.

**phase** The number of separate field winding circuits.

**rotor** The internal part of the stepper motor that rotates.

**single-step mode** Operating the motor at a slow enough rate so that it can be stopped after any step without overshooting.

**slew mode** Stepping the motor at a faster rate than the single-step mode; used to move to a new position quickly. The motor will overshoot if the speed is not ramped up or down slowly.

**stalling** A situation wherein the motor cannot rotate because the load torque is too great.

**stator** The part of a stepper motor that surrounds the rotor and consists of field poles (electromagnets).

**stepper motor** A motor that rotates in steps of a fixed number of degrees each time it is activated.

**three-phase stepper motor** A motor with three separate sets of field coils; usually found with VR motors.

**two-phase stepper motor** A motor with two field circuits. This motor requires polarity reversals to operate and hence is bipolar.

**unipolar motor** A motor that does not require polarity reversals. A four-phase motor is unipolar.

**variable-reluctance (VR) stepper motor** A motor that uses a toothed iron wheel for the rotor and consequently can take smaller steps but has no detent torque.



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## EXERCISES

### Section 8.1

1. A  $15^\circ$  stepper motor is commanded to go 100 steps CW and 30 steps CCW from a reference point. What is its final angle?
2. A  $7.5^\circ$  stepper motor is commanded to go 50 steps CCW, 27 steps CW, and 35 steps CCW again. What is its final angle (referenced from its original position)?
3. Why can a stepper motor be operated open-loop in a control system?
4. What is the detent (or residual) torque in a PM stepper motor, and what causes it?
5. A stepper motor is being used as a crane motor in an expensive toy. The motor has the following properties: Holding torque = 35 in. · oz, dynamic torque = 20 in. · oz, and detent torque = 5 in. · oz. The motor turns a 1.5-in. diameter pulley around which a string is wound. How much weight can the crane lift? How much weight can the crane continue to support with the power on; with the power off?
6. List the stepping sequence you would use to make the two-phase motor of Figure 8.8 rotate CW.
7. List the stepping sequence you would use to make the two-phase motor of Figure 8.9 operate as an eight-step drive (CCW).
8. List the stepping sequence you would use to make the four-phase motor of Figure 8.12(a) operate as an eight-step drive (CCW).

### Section 8.2

9. Explain the principle of operation of a VR stepper motor.
10. Does a VR stepper motor have a detent torque? Explain.

### Section 8.3

11. Explain the principle of operation of a hybrid stepper motor.
12. What are the advantages of the hybrid stepper motor?

### Section 8.4

13. A 5-V stepper motor is to be microstepped with one-tenth steps. List the voltage table required for this [similar to Figure 8.27(b)].
14. What is the purpose of a ballast resistor on a stepper motor drive, and how does it work?
15. What is the purpose of bilevel drive, and how does it work?
16. How does a chopper drive improve torque at higher stepping rates?

### Section 8.5

17. A  $1.8^\circ$  stepper motor turns a leadscrew that has 24 threads per inch.
  - a. How many steps will it take to advance the leadscrew 1.25 in.?
  - b. What is the linear distance the leadscrew advances for each step?

- 18.** A  $7.5^\circ$  stepper motor (four phase), controlled by a computer, is used to position a telescope through a gear train. The telescope must be positioned to within  $0.01^\circ$ . A front panel has toggle switches that are used to specify how far the telescope is to move (LSB =  $0.01^\circ$ ). The total range of the telescope is  $0$ - $60^\circ$ .
- How many toggle switches would be required?
  - What gear ratio would you specify?
  - Draw a block diagram of the system, showing all parts of the system and specifying the gear ratio.

# CHAPTER 9

## Alternating Current Motors

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### OBJECTIVES

After studying this chapter, you should be able to:

- Describe the characteristics of single-phase AC and three-phase AC motors.
- Understand the principles of operation of the AC induction motor.
- Explain how single-phase induction motors are started, including the concept of run and start windings and start capacitors.
- Explain how to reverse the two-phase (split-phase) motor.
- Understand the principles of operation of the synchronous motor.
- Describe the concept of power factor and power-factor correction.
- Understand the principles of AC motor control, including start-stop control, jogging, reduced voltage starting, and variable-speed control with a DC link converter.

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### INTRODUCTION

In terms of sheer numbers, the AC induction motor (Figure 9.1) is the most widely used type of electric motor in the modern world. AC motors are primarily used as a source of constant-speed mechanical power but are increasingly being used in variable speed-control applications. They are popular because they can provide rotary power with high efficiency, low maintenance, and exceptional reliability—all at relatively low cost. These desirable qualities are the result of two factors: (1) AC motors can use the AC power “right off the lines.”—DC motors require the added expense of a rectifier circuit; (2) most AC motors do not need brushes as DC motors do. In most cases, the AC power is connected only to the motor’s stationary field windings. The rotor gets its power by electromagnetic induction, a process that does not require physical electrical contact. Maintenance is reduced because brushes do not have to be periodically replaced. Also, the motor tends to be more reliable and last longer because there are fewer parts to go wrong and there is no “brush dust” to contaminate the bearings or windings.

**Figure 9.1**

An AC induction motor.



Despite these advantages, there is a problem with using AC motors in control systems: These high-efficiency AC motors are by nature constant speed, and control systems usually require the motor speed to be controllable. As you recall, the speed of a DC motor can be controlled by simply adjusting the applied voltage. For complete speed control of an AC motor, both voltage and frequency must be adjusted, which requires using special electronic speed-control circuitry, such as the volts-per-hertz (V/Hz) drive or the vector drive (discussed later in this chapter).

Still, the most common use of AC motors is for the many applications where speed control is not necessary. Included here are fans, pumps, mixers, machine tools, hydraulic power supplies, and household appliances, to name but a few.

Like DC motors, AC motors come in all sizes. **Integral-horsepower (HP) motors** are those with a power rating of 1 hp to over 1000 hp, and **fractional-horsepower (HP) motors** are those rated less than 1 hp. Typically, the larger motors use three-phase AC power with voltages from 208 to 600 Vac, and the smaller motors use single-phase AC power, with either 120 Vac or 240 Vac.

There are a number of different types of AC motors. The most common by far is the general-purpose induction motor, which is the familiar type found in major appliances and in machine shops. Other types of AC motors include the synchronous motor, the AC servomotor, and the universal motor.

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## 9.1 AC POWER

### Background

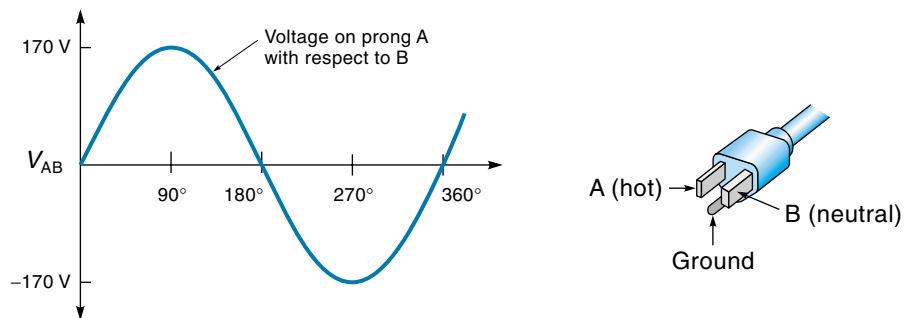
Thomas Edison proposed the first power-distribution system, which was to be a DC system. A major drawback of this proposed system was that the power consumer had to be within 10 miles or so of the generating station because of the significant power losses

along the wires. Then in the early 1900s, Nikola Tesla invented the AC motor and generator, and with George Westinghouse built a large AC-generating station at Niagara Falls, New York. This system demonstrated the tremendous advantage of using AC for power distribution, namely, much reduced line losses. This is achieved by using a transformer at the generating station to increase the voltage and reduce the current. Thus modified, the power can be transmitted through smaller wires with much less power loss. At the destination site, another transformer reduces the voltage back to usable levels (and increases the current). Because transformers work only with AC, the AC system was destined to become the standard commercial form for transmitting electric power.

### Single-Phase AC

Before discussing how AC motors work, let's briefly review exactly what AC power is. The simplest form of AC, and the kind most people are familiar with, is **single-phase AC**, the standard diagram of which is shown in Figure 9.2(a). How does this sine-wave diagram relate to the actual instantaneous voltages that would be found on prongs A and B of a standard power plug [Figure 9.2(a)]? To really understand AC, we must go back to the original definition of voltage, which is the difference in potential between two points (notice it says nothing about *ground*). Therefore, the sine wave in Figure 9.2(a) is showing only the voltage difference between prongs A and B for one complete cycle. For example, in the first half of the cycle, prong A is more positive than B by up to 170 V. At exactly midcycle ( $180^\circ$ ), there is no voltage difference between A and B. During the second half of the cycle, prong A has a negative voltage with respect to B. The power company maintains the voltage of prong B at ground, which means it is not only the return path for the current but also is actually connected to a metal rod driven into the earth. Therefore, the wire that goes to the wide prong (B) is called the **neutral, return wire** or *cold side* (usually white in color) and is theoretically safe to touch. (However, this is not recommended because it could be wired wrong!) The wire going to prong A is called the **hot side** because it carries the voltage (usually black in color and should definitely *not* be touched!). The

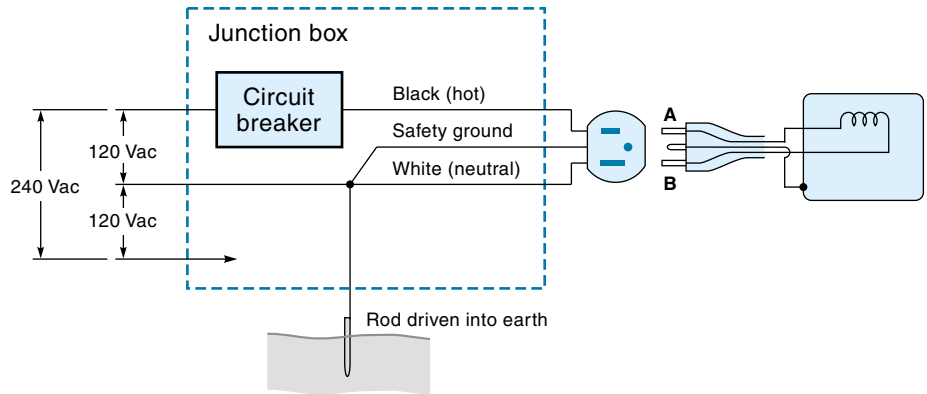
**Figure 9.2**  
120/240 Vac single-phase power.



(a) 120 Vac waveform and plug

Figure 9.2

(cont.)



(b) 120 Vac power-distribution circuit

third pin on the plug [Figure 9.2(b)] is the *safety ground* and is kept separate from the neutral wire until the actual grounding point. This ground wire is typically connected to the chassis or motor frame, which prevents you from getting a shock from touching the equipment because it is at the same potential as the earth.

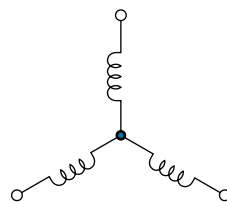
The highest instantaneous voltage on (prong A) is 170 V (called  $V_{\text{PEAK}}$ ) and is considerably higher than the AC designation of 120 Vac (called  $V_{\text{RMS}}$ ) because the number 120 represents the effective voltage if the power content were constant over the whole cycle. Single-phase AC is supplied as 120 Vac and 240 Vac [usually combined together on three wires as shown in Figure 9.2(b)]. Fractional hp motors are commonly 120 Vac, and larger motors usually use 240 Vac. There are three common line frequencies in use—60 Hz in the United States and Canada, 50 Hz in most of the rest of the world, and 400 hz on many aircraft and shipboard systems. (The higher frequency reduces the amount of iron required in transformers and motors and thus reduces the weight.)

### Three-Phase AC

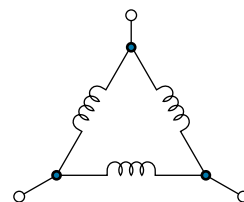
Single-phase AC is the most common form of power in homes and businesses, but it is not the best type of AC for transmitting power or running AC motors. That distinction

Figure 9.3

Two types of connections for three-phase AC.



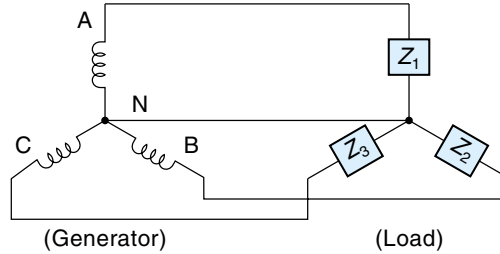
(a) Wye-connection



(b) Delta-connection

**Figure 9.4**

A three-phase power system (wye-connection).



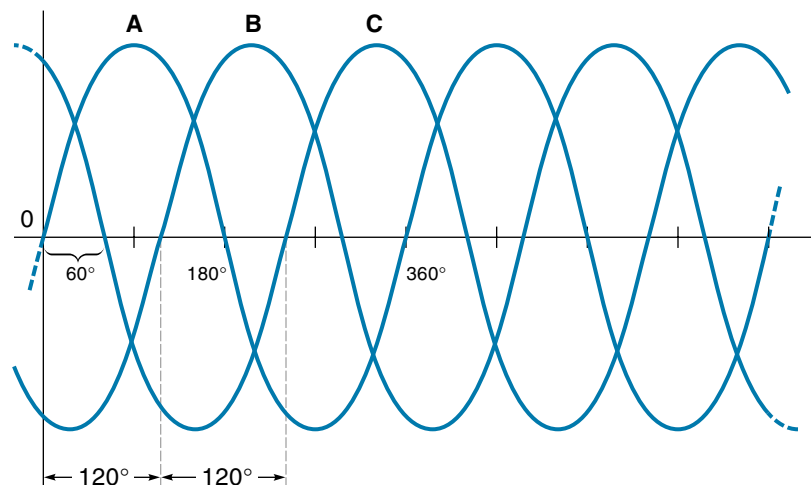
belongs to **three-phase AC**, which is more complicated; however, because of its advantages, it is used extensively by heavy power users.

Three-phase power is created by a three-phase generator that has three separate field windings spaced  $120^\circ$  apart. Two configurations are possible: the **wye-connection** (or Y-connection), shown in Figure 9.3(a), and the **delta-connection** (or  $\Delta$ -connection), shown in Figure 9.3(b). A schematic of a wye-connection generator is shown in the left side of Figure 9.4. One end of each coil is connected to a common wire labeled N (neutral), and the other ends of the coils are labeled A, B, and C, which are the three-phase outputs. The relationship of **phase voltages** A, B, and C is presented in Figure 9.5, which shows each phase voltage with respect to neutral. Each phase makes a complete cycle in one period and is  $120^\circ$  apart; at any particular time, all phases add to 0 V (for example, add the magnitudes of the three voltages when  $t = 0$ ). Notice in Figure 9.5 that the **phase sequence** is ABC (first A, then B, then C). The other possible sequence would be ACB.

Typically, three-phase power is distributed within a facility as wye-connected. Figure 9.4 shows a complete wye-connected system where a three-phase generator is connected to a three-phase load (such as a three-phase motor). If all the load impedances ( $Z_1$ ,  $Z_2$ , and  $Z_3$ ) are the same, then the system is said to be balanced, and there would actually be no current in the neutral wire. Under these conditions, the neutral wire could be removed, and the motor

**Figure 9.5**

Three-phase AC waveforms.



would continue to run just fine. In real life, it is unlikely that any particular facility would present a purely balanced load to the power lines, because each phase voltage, if considered separately, can be used as a standard single-phase voltage. Therefore, a single-phase load such as a light bulb could be (and frequently is) connected between one of the phase voltages and neutral. The system would now be unbalanced, and some current will flow in the neutral wire. In practice, effort is made to keep the load on each phase equal. Finally, the specified **line voltage** of a three-phase wye-connected system is the vector sum of two individual phase voltages in series. This relationship is given in Equation 9.1:

$$\text{For wye-connected: } V_{\text{Line}} = \sqrt{3} V_{\text{Phase}} \quad (9.1)$$

$V_{\text{Line}}$  = three-phase line voltage (voltage between any two lines)

$V_{\text{Phase}}$  = individual phase voltage (voltage across one coil, which for a wye system is also the voltage between a line and neutral)

For a delta-connected system, the line and phase voltages are the same.

### EXAMPLE 9.1

A three-phase wye-connected system has individual phase voltages of 120 Vac. Find the three-phase line voltage.

### SOLUTION

Using Equation 9.1,

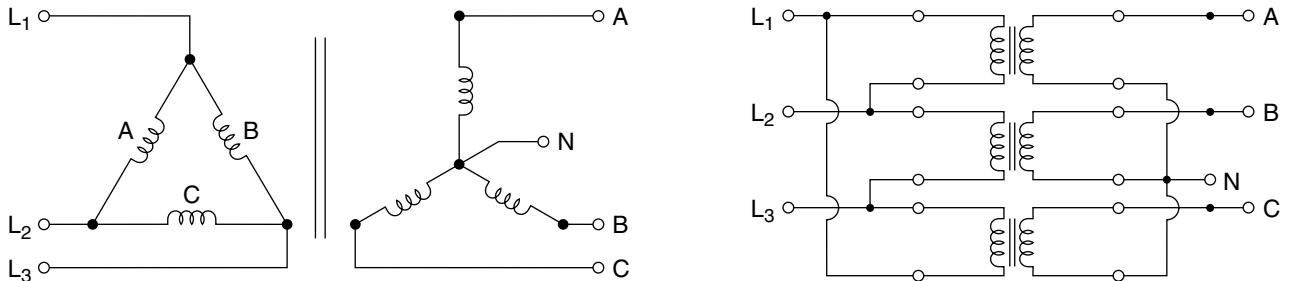
$$V_{\text{Line}} = \sqrt{3} V_{\text{Phase}} = 1.73 \times 120 \text{ Vac} = 208 \text{ Vac}$$

Wye-connected three-phase power is mainly used for distribution within a facility because the neutral wire allows separate single-phase loads to be pulled off as needed. Delta-connected systems have no neutral wire and are inherently balanced. The power company typically uses three wires to transmit the three-phase power over the power lines. Once at the site the power is easily converted to wye-connected power with special transformers designed for that purpose. Three-phase transformers can be made as one unit or from three single-phase transformers (see Figure 9.6).

### Electrical Safety

Electricity can injure, permanently disfigure, or kill human beings. Strickly speaking, it's not the voltage but the current that does the damage. Large currents, above 2 amps, literally cook the flesh, causing severe burns that heal slowly if at all. Currents between 1 and 2 amps damage nerve centers, can cause temporary or permanent paralysis, and may be lethal if the heart is involved. A current in the range of 100–200 mA causes the muscles to flutter uncontrollably; here again, if the heart is involved, it stops working





(a) True three-phase delta-wye transformer

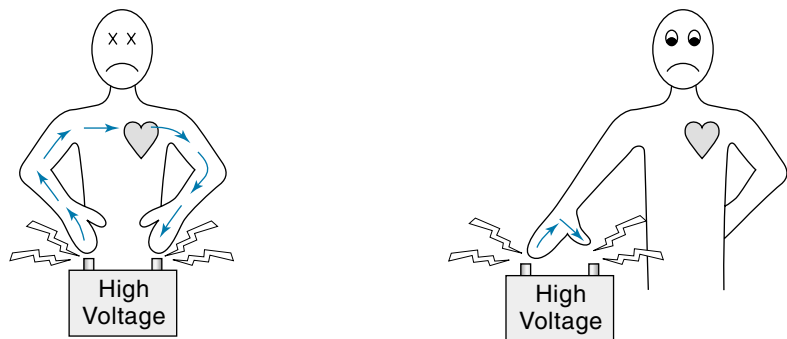
(b) Three single-phase transformers connected as a delta-wye

**Figure 9.6**

Three-phase delta-wye transformers.

properly. This is probably the most common form of death by electrocution. Currents from 25 to 75 mA may cause unconsciousness. This may actually save the unconscious person, who may collapse and so disengage. Currents around 25 mA tend to make the muscles contract and “lock up” (a horrible situation because the victim can’t let go). Even smaller-current shocks are dangerous, because they can cause a person to jump or lurch violently and perhaps hit something or fall.

How does current enter the body? By touching two different electrical potentials at the same time (usually the high voltage and ground). In such a case, the body completes the electric circuit and current flows. The amount of current depends on the voltage and the resistance of the body. The resistance of dry skin is high enough so that a person doesn’t even start to feel a shock until around 50 V; but by 120 V a painful shock is felt. Wet skin offers less electrical resistance, so any voltage delivers more of a shock if the skin is wet. Two examples of a person being shocked are shown in Figure 9.7. In Figure 9.7(a) the man is touching the voltages with both hands, so the current path goes near his heart—a very dangerous situation. If you must work around high voltages, it



(a) When voltage is touched with both hands, current goes through the heart

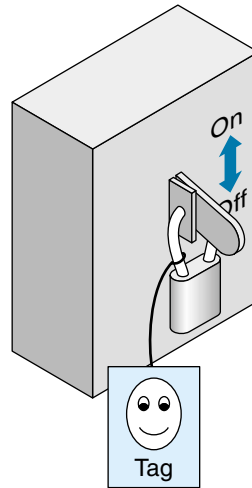
(b) When voltage is touched with one hand, current is away from the heart

**Figure 9.7**

Issues in electrical safety.

Figure 9.7

(cont.)



(c) Switch box that can be padlocked in the off position

is much safer to use only one hand as shown in Figure 9.7(b), because the probable current path is then far away from the heart.

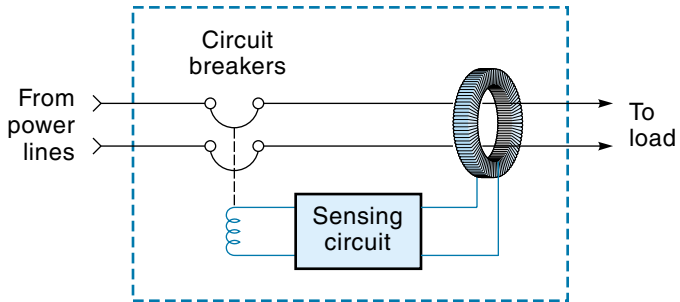
Electrical safety depends on two things: safe equipment and safe practices. Safe equipment means that all high-voltage surfaces are covered in some way to prevent accidental contact. In high-voltage areas, placing rubber mats on the floor and using of rubber gloves and nonconductive fiberglass ladders (if needed) will reduce the possibility of personnel getting shocked. All chassis, cases, and surrounding metal (as well as the neutral power line) should be well connected to earth ground. This promotes safety, because if a high-voltage line got loose and somehow touched the case (or developed a “short” to the case), the resulting current would be dumped to ground, but the case itself would stay at a safe ground voltage (as illustrated back in Figure 3.43).

People who work around high voltages must follow some basic safety rules:

1. *Always turn off the power before you work on the equipment.*
2. If equipment can't be unplugged (for example, a huge printing press occupying many square feet), then be sure its power feed line goes through a switch box that can be padlocked in the off position [see Figure 9.7(c)] Before you work on the equipment, you would lock the switch in the off position and put a paper tag on the switch stating who you are and letting people know that you are in there working on the equipment. When you are finished working, you remove your tag and lock. No one should be allowed to remove it but you (under normal circumstances).
3. Even if you have turned the power off at the master switch, there is always the possibility that you locked the wrong switch or that it is miss-wired. To be sure, you should actually measure the high-voltage line with a meter (or two different meters) to confirm that the power really is off.

**Figure 9.8**

A ground-fault interrupter for life protection.



4. Once you have verified that the power is off, you can temporarily jumper the high-voltage line to ground, which will protect you should someone, somehow, turn the power back on before you are finished. (However, this practice is appropriate only if there are protective relays in the line that will open when there is a short circuit.)

### Ground-Fault Interrupters

The principle of a **ground-fault interrupter** (GFI) is to disconnect the power if a current leakage to ground is detected. There are two types of GFIs: those designed for *life protection* and those designed for *equipment protection*.

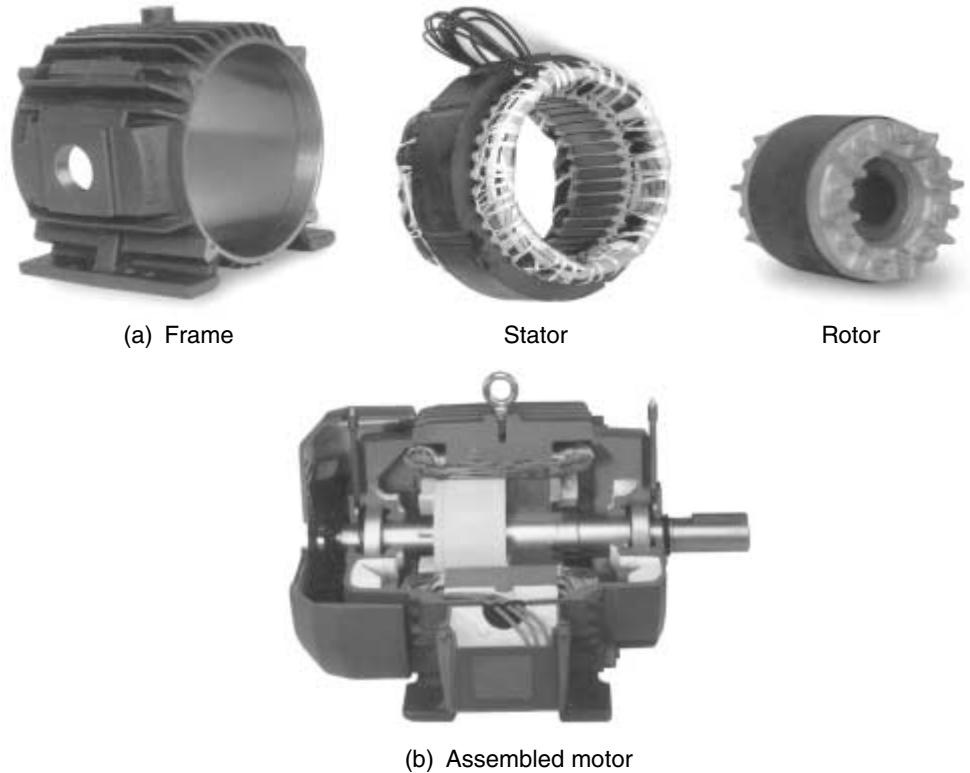
A GFI used for *life protection* is to prevent shock hazard, particularly in bathrooms and around water. Without a GFI, a person who touches a hot wire while standing in water or on a wet floor will get shocked because his or her body provides an electrical path to ground. The GFI circuit opens a circuit breaker quickly if it senses that more current is in the hot wire than in the neutral wire, indicating that the current has found an alternate path to ground—that is, through someone’s body. Figure 9.8 shows a GFI used for life protection. Notice that power-line conductors pass through a toroidal coil. If this coil senses any (current) unbalance, a solid-state circuit directs a circuit breaker to open.

A GFI used for *equipment protection* is to prevent equipment damage and possibly fire due to internal arcing. Under normal circumstances, a current should not flow in the safety ground wire. For example, in Figure 9.2(b), the motor current should flow only through the hot and neutral wires, not in the ground wire. Only if there is a short circuit or fault between the hot wire and the motor chassis will there be a current in the ground wire. The fault might not draw enough current to trigger the overload device, but nonetheless could be a fire hazard. The GFI monitors the current in the ground wire and opens a circuit breaker if the current exceeds a preset level.

## 9.2 INDUCTION MOTORS

By far the most commonly used type of AC motor is the induction motor, the simple, reliable, “workhorse” that powers most domestic and industrial machines. The basic parts of

**Figure 9.9**  
An induction motor.



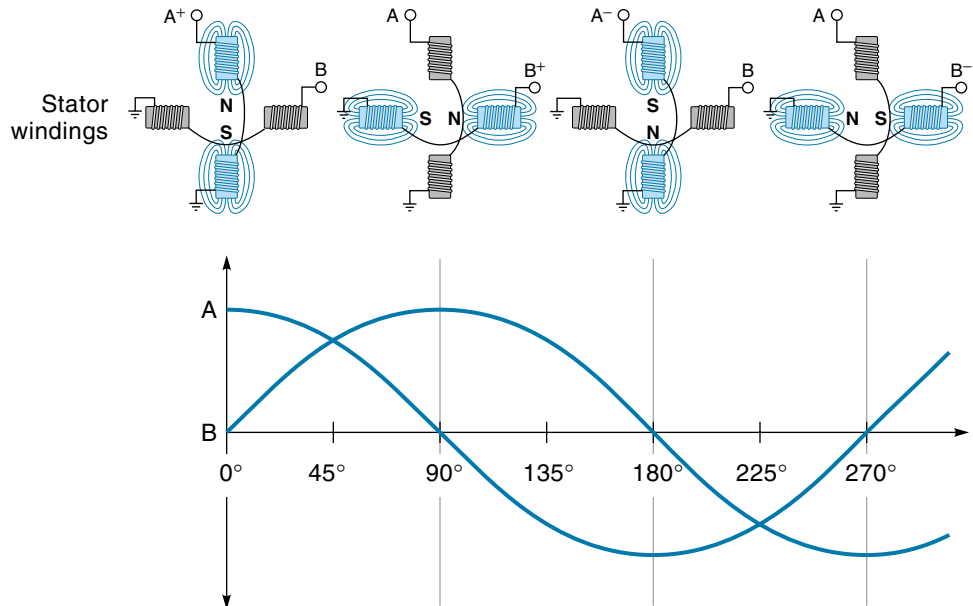
the induction motor are the **frame**, **stator**, and **rotor** [Figure 9.9(a)]. The stator consists of the stationary field coil windings. The rotor is positioned inside the stator and rotates as a result of electromagnetic interaction with the stator. The frame supports the stator and rotor in the proper position shown in [Figure 9.9(b)].

### Theory of Operation

The theory of operation of the AC induction motor has some similarities to that of the stepper motor or BLDC (brushless DC motor). Recall that those motors work by having their field poles energized in sequence around the stator. The rotor is pulled around because it is attracted to the sequentially energized poles. With stepper motors and BLDCs, special switching circuits are required to turn the field windings on and off. The AC motor also works by rotating the stator field, but it makes use of the natural alternating nature of the AC wave to turn the field coils on and off sequentially. The AC induction motor does not need brushes because the rotor is essentially a passive device that is continuously being pulled in one direction. To use an old analogy, the rotor is the “horse,” and the rotating stator field is the “carrot.”

**Figure 9.10**

How two-phase AC causes a rotating field.



To explain the principles of how the AC wave can be used to sequentially energize the field coils, we will examine the operation of a theoretical two-phase motor. Two-phase AC consists of two individual phase voltages (Figure 9.10). Notice that phase B is lagging behind phase A by  $90^\circ$ —that is, phase A peaks at  $0^\circ$ , and phase B peaks  $90^\circ$  later. The two-phase motor (Figure 9.11) is connected so that phase A energizes the top and bottom poles and phase B energizes the left and right poles. The action of two-phase AC on the motor is to cause the stator magnetic field to effectively rotate clockwise (called a **rotating field**), even though the coils themselves are stationary.

In Figure 9.10, at  $0^\circ$  phase A is at peak voltage while phase B is 0 V. At this point, phase A has all the voltage, and phase B has none; therefore, the windings connected to phase A (top and bottom) will be energized, and the windings connected to phase B (left and right) will be off. This situation is depicted in the coil drawing (upper left) of Figure 9.10. The polarity of the applied voltage causes the top winding to present a north (N) magnetic pole to the rotor and the bottom winding to present a south (S) magnetic pole to the rotor.

At  $90^\circ$  later in the power cycle (Figure 9.10), phase A voltage has gone to 0 V (deenergizing the top and bottom windings), and phase B has risen to peak voltage, energizing the left and right windings. Specifically, the positive phase B voltage will cause the right-side winding to present a north magnetic pole to the rotor and the left winding to present a south magnetic pole (as indicated in the coil drawing of Figure 9.10).

At 180°, phase B voltage has gone back to 0 V (deenergizing the left and right windings), and phase A has descended to a negative peak voltage. Once again, the top and bottom windings are energized but this time with the opposite polarity from what they were at 0°, causing the magnetic poles to be reversed. Now the bottom winding presents a north magnetic pole to the rotor, and the top winding presents a south magnetic pole.

At 270°, phase A has ascended to 0 V (deenergizing the top and bottom windings), and phase B has gone to a negative peak. Once again, the left and right windings are energized but this time with the left winding presenting a north magnetic pole to the rotor and the right winding a south magnetic pole.

This analysis explains how two-phase AC\* causes the magnetic field to act as if it were rotating in a clockwise (CW) direction. (You can see this in the coil drawings of Figure 9.10, where the north pole apparently rotates CW.) What was not apparent from the discussion is that the rotation of the field is smooth and continuous—it doesn't jump from pole to pole as might be inferred from the discussion. For example, consider the situation at 45°. From Figure 9.10, you can see that both sets of poles are partially energized, causing the resultant N-S magnetic field to be halfway between the two poles.

Mechanical rotation of the rotor is the result of the rotor being pulled around in a CW direction, “chasing” the rotating field. (The rotor is magnetized by an “induced” current from the field—a concept that will be explained later.) The field makes one complete revolution per cycle; thus, for a line frequency of 60 Hz, the field would rotate at 3600 rpm—(60 cycle/s) × (60 s/min) = 3600 rpm. The speed of the rotating field (3600 rpm in this case) is called the **synchronous speed**. For an induction motor, the rotor speed does not exactly match the synchronous speed, it's slightly lower.

The motor under discussion (Figure 9.11) would be called a *two-pole motor*, even though it has four individual poles, because the “pole” designation of an AC motor refers to the number of individual *poles per phase*. The reason? Motor speed is proportional to poles per phase, not simply the number of poles. In this case, 4 poles/ 2 phase = two-pole. An AC motor can be built to go slower than 3600 rpm by increasing the number of pole sets. For example, if the motor in Figure 9.11 had two complete sets of poles, it would turn one-half revolution for each cycle, giving it a synchronous speed of 1800 rpm. (This would be called a *four-pole motor*, because 8 poles/2 phase = four-pole.) If the motor had three sets of poles (making it a *six-pole motor*), it would turn at a synchronous speed of 1200 rpm (which is one third of 3600 rpm). This idea can be expressed in a synchronous equation:

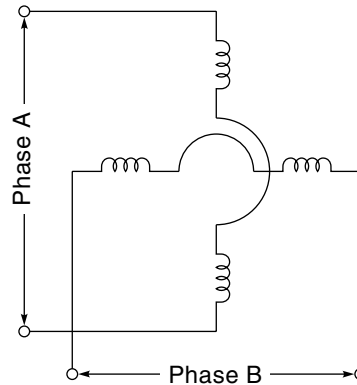
$$S_s = \frac{120f}{P} \quad (9.2)$$

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\*Power companies do not supply two-phase power, only single-phase and three-phase. However, two-phase power can be “created” from single-phase power using a capacitor and is used in some special-purpose AC motors. This discussion is based on two-phase power because I feel it is the best way to explain the concept of a rotating field.

**Figure 9.11**

A two-phase motor.



where

$S_s$  = synchronous speed in rpm (motor speed is actually somewhat less, as will be explained later)

$f$  = frequency of the AC line

$P$  = number of field poles per phase

### EXAMPLE 9.2

- Find the synchronous speed of a 60-Hz, four-pole, single-phase motor.
- How many individual poles does this motor have?

### SOLUTION

- For a four-pole motor,  $P = 4$  in Equation 9.2:

$$S_s = \frac{120 \times 60}{4} = 1800 \text{ rpm}$$

which is the synchronous speed.

- To find the number of individual poles, recognize that this motor has four poles per phase and one phase; therefore, it has four poles.

### EXAMPLE 9.3

- Find the synchronous speed of a 60-Hz, four-pole, three-phase motor.
- How many individual poles does this motor have?

**SOLUTION**

- a. For a four-pole motor,  $P = 4$  in Equation 9.2:

$$S_S = \frac{120 \times 60}{4} = 1800 \text{ rpm}$$

- b. To find the number of individual poles, recognize that this motor has four poles per phase and three phases; therefore, it has 12 poles.

Some motors are designed so that the number of poles can be changed by reterminating wires in a junction box. This way the same motor can be set up to run at different speeds as the application requires; for example, when it's operating as a two-pole motor, the speed is 3600 rpm, and when it's operating as a four-pole motor, the speed is 1800 rpm.

The rotor of the AC motor must act like a magnet in order for it to be pulled around by the rotating stator field. A diagram of a rotor is presented in Figure 9.12(a). It consists of a number of aluminum or copper bars connected with two end rings. Because this configuration reminded someone of a squirrel cage, it is called a **squirrel cage rotor**.

The squirrel cage rotor has no magnetic properties when the power is off. However, when AC power is applied to the stator windings and the stator field starts rotating, the rotor becomes magnetized by induction. It works like this: As the stator field rotates past an individual bar, the field strength in the bar rises and falls. This changing magnetic field induces a voltage in the bar, and the voltage causes a current to flow. The current flows through the bar, through the end rings, and back through other bars. This current causes the bar to have a magnetic field, and it is this field, interacting with the rotating stator field, that produces the mechanical torque. Figure 9.12 (b) shows a rotor with bars that are actually embedded in slots of a ferromagnetic core. No insulation is between the bars and the iron core because the bars present a much lower resistance than the laminated iron sheets that make up the core.

For this motor to work, a voltage must be induced from the stator to the rotor, but this can only happen when the magnetic field is changing. As far as the rotor is concerned, the changing field occurs as a result of the stator field rotating past it. This means that, even though the rotor is chasing the rotating stator field, it can never quite catch up to the synchronous speed; if it did, no relative motion would be between the rotor and stator and consequently no induced voltage in the rotor. Thus, the rotor is always going slower than the rotating field, and this difference is referred to as **slip**. Slip is frequently given as a percentage (usually less than 10%) of the stator speed as described in the following equation:

$$\text{Slip} = \frac{S_S - S_R}{S_S} \quad (9.3)$$

where

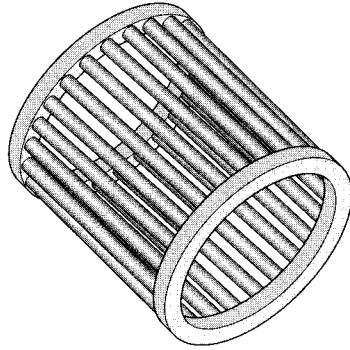
$S_S$  = synchronous (stator) speed (in rpm)

$S_R$  = rotor speed (rpm)



**Figure 9.12**

An induction motor “squirrel cage” rotor.



(a) Diagram of “squirrel cage”



(b) Assembled rotor

**EXAMPLE 9.4**

A typical washing machine motor has a rated speed of 1725 rpm. What is the slip in percentage at this speed?

**SOLUTION**

Because it is a 60-Hz motor, we know that the stator speed must be a simple fraction ( $1/2$ ,  $1/3$ ,  $1/4$ , and so on) of 3600 rpm. Quick calculations reveal that  $1/2$  of 3600 is 1800, which is slightly above 1725 rpm. Therefore, 1800 must be the synchronous speed. Now we use Equation 9.3 to find the slip:

$$\frac{S_S - S_R}{S_S} = \frac{1800 - 1725}{1800} = 0.042 = \mathbf{4.2\%}$$

Some speed control of the AC motor is possible by controlling the amount of slip by varying the voltage. For a constant load, the slip will increase as the voltage is lowered because, reducing the applied voltage reduces the strength of the “rotating” magnetic field, and the rotor must slip more to get its required induced current. This type of control can be used to change the speed 10-15% or more, depending on the rotor design.

Induction motors are classified by the National Electrical Manufacturers Association (NEMA). The most common type, NEMA class B, has a low-resistance rotor, which allows

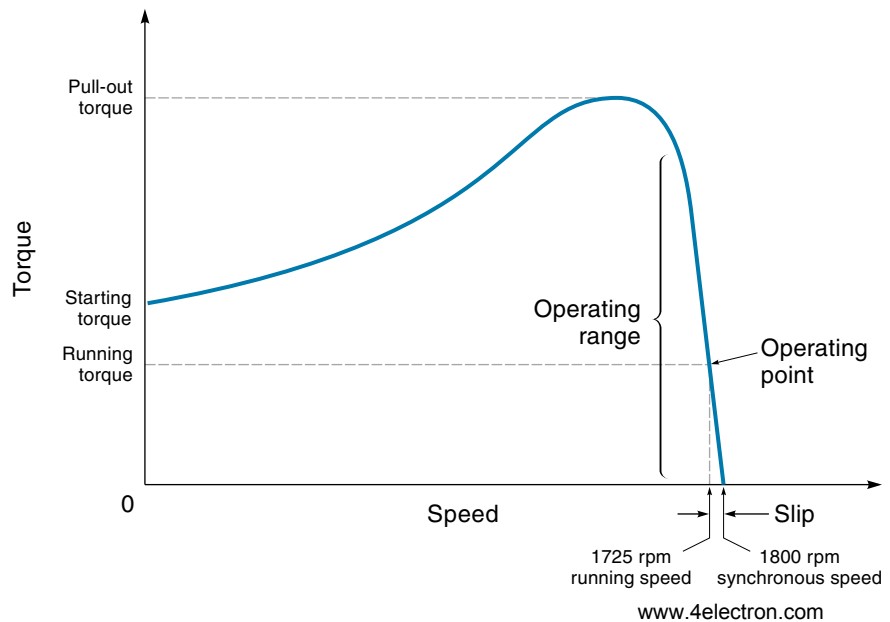
it to operate with low slip and high efficiency. Another type, NEMA class D, has a high-resistance rotor, which allows it to operate with more slip and consequently more torque.

A torque-speed curve for the motor of Example 9.4 is shown in Figure 9.13, and it is very useful for understanding motor behavior. This curve shows the torque available at various motor speeds. The maximum possible speed (1800 rpm in this case) is the speed of the rotating stator field, that is, the synchronous speed. However, there is no torque at this speed because the motor needs some slip to operate. To get full-load torque, a point defined by the manufacturer, the rotor must be slipping some—in this case, reducing the speed to 1725 rpm. These motors have good **speed regulation** in the operating range—that is, the speed changes little over a wide range of load torque demands. Also in Figure 9.13, notice that the AC motor behaves something like the DC motor in the operating range—that is, as the load increases the speed decreases. However, at a certain point called the **pull-out torque**, the torque drops off sharply, even though the speed is still decreasing. In real life, this means that if the load on a motor was increasing, the speed would decrease until the motor load reached the pull-out torque, at which point the motor would quickly **stall** (stop turning). The **starting torque** will depend on the type of motor, but it is typically 1.5 times greater than the rated load, and unlike the DC motor, it is not the maximum torque available.

## Single-Phase Motors

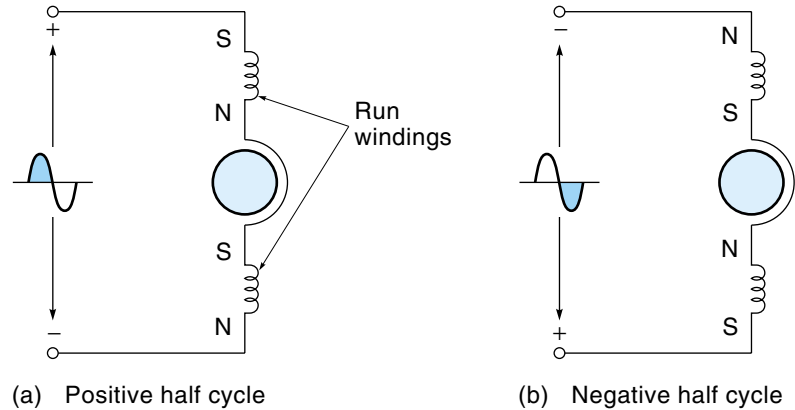
AC induction motors are grouped into classifications depending on the type of power they use: single-phase or three-phase (two-phase motors actually are supplied with single-

**Figure 9.13**  
Torque-speed  
curve of an  
induction motor.



**Figure 9.14**

A single-phase motor showing how the field reverses with each half-cycle of AC.



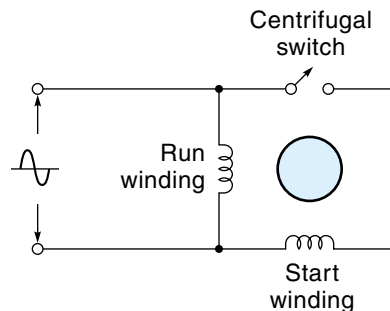
phase power). Each type is based on the same operating principles presented previously in this chapter, and each type has its advantages and disadvantages.

Single-phase AC is the most common form of power and is available almost everywhere. For this reason, the single-phase motor is in wide use for domestic and commercial purposes, typically for applications requiring up to 1 or 2 hp. The basic operational theory is the same as for the motor discussed before, but this motor has one major problem: It can't start itself. To understand this, look at the single-phase motor diagram shown in Figure 9.14. This motor has only one set of stator windings (called the **run windings**), which are positioned across from each other. As the AC voltage cycles from positive to negative, the magnetic field reverses itself [Figures 9.14(a) and (b)]. These field reversals cannot truly be said to be a rotating field, so they cannot provide a starting torque to the rotor. They can, however, keep the rotor turning if it's started in some other way. The trick, then, is to provide a way to start the single-phase motor.

The most common way to start a single-phase motor is to use a second set of windings, called the **start windings**, which are only energized during the start-up period. In effect, the motor temporarily becomes a two-phase motor, which *is* self-starting. This type of motor is known by the name **split-phase**. The AC in the start winding should ideally be  $90^\circ$  out of phase with the run winding. The simple split-phase motor shown in Figure 9.15 relies on the fact that the start winding, which consists of a few turns of

**Figure 9.15**

A single-phase motor (split-phase).

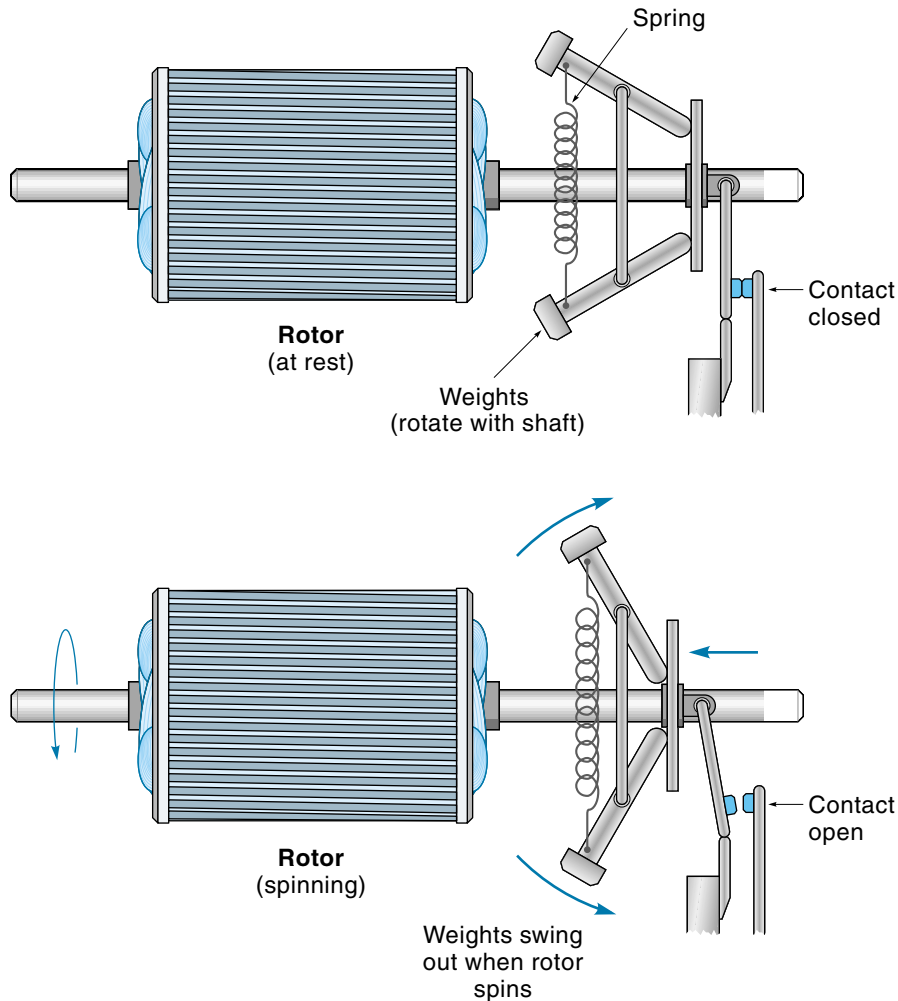


thin wire, has much less inductance and much more resistance than the run windings and so creates a phase shift with respect to the run winding. When the motor is up to about 80% speed, a switch opens automatically by centrifugal force (Figure 9.16), disconnecting the start windings. The motor then continues running on only the run windings. This starting method provides about 40-50° of phase shift, enough to start the motor but not enough to provide a lot of start-up torque.

Another way to create the necessary phase shift between the run and start windings is with a nonpolarized capacitor. This motor, shown in Figure 9.17, is a **capacitor start motor**. When the motor is started, the centrifugal switch is closed, and the capac-

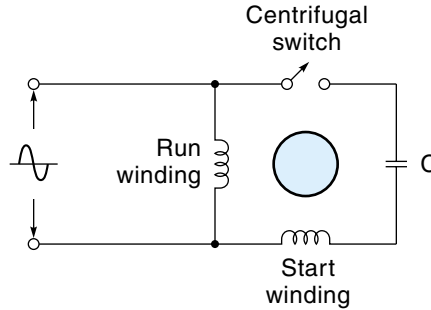
**Figure 9.16**

A centrifugal switch disconnects the start winding when the rotor is up to speed.



**Figure 9.17**

A single-phase motor (split-phase) using a capacitor with a centrifugal switch.



itor causes the current in the start winding to lead the current in the run winding. Once the motor is running, the switch opens, and the motor continues to operate in single-phase mode on the run windings. The capacitor start motor can achieve a full  $90^\circ$  phase shift between run and start windings, with the consequence of a high starting torque.

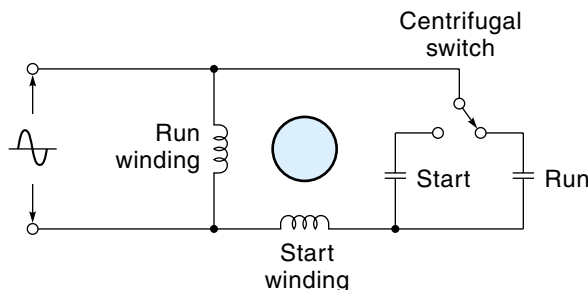
There are a number of other variations of the split-phase type of the single-phase motor. In one design, called the **permanent-split capacitor motor**, the start windings (with capacitor) are permanently connected, eliminating the need for the centrifugal switch. This design increases the torque of the motor, simplifies the hardware, increases the reliability, and improves the power factor (power factor is discussed later in the chapter). These motors also run more quietly with less vibration. Another motor design, the **two-capacitor motor**, is shown in Figure 9.18. This motor uses a larger value capacitor for starting and then switches to a lower value for running. Two-capacitor motors enjoy the same performance benefits as the permanent split-capacitor motor, because the run capacitor introduces some phase shift, which (as shown in Figure 9.10) more closely approximates a rotating field.

All single-phase motors discussed so far will start off turning in the same direction with each power up. By reversing the polarity of one of the windings, the motor will start and run in the opposite direction.

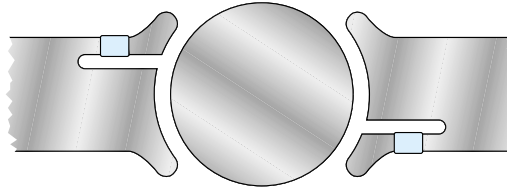
Fractional hp motors can be **shaded-pole motors**, which do not require a separate start winding for them to start. Instead, a small portion of each pole is separated from

**Figure 9.18**

A two-capacitor motor.



**Figure 9.19**  
A shaded-pole motor.



the rest and encircled by a copper band, as shown in Figure 9.19. When power is applied to the motor, a current is induced in this copper band, which results in the magnetic field being delayed in its vicinity. This unbalance creates a weak rotating component to the stator field, which is enough to start the motor turning. This motor design is simple and easy to construct but is inefficient and has low torque. Consequently, it is usually used where only low power and a nearly constant load is required, such as small blowers.

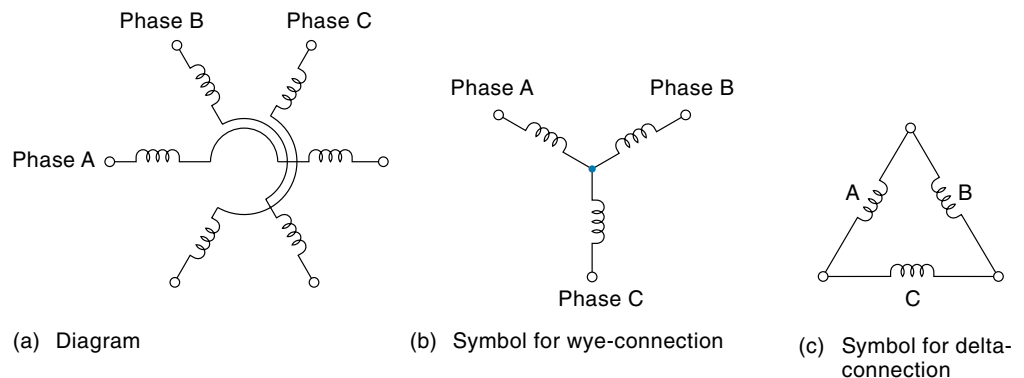
### Three-Phase Motors

The three-phase motor is simpler and smaller than its single-phase counterpart, but it can be used only where three-phase power is available. Usually, this is an industrial site; therefore, these motors tend to be from 2 hp up. Figure 9.20 shows a diagram of the three-phase motor (notice that it can be either a wye or a delta connection). It has three sets of stator windings, with each set of windings being powered by one of the phase voltages. The natural timing sequence of the three individual phase voltages produces the rotating stator field that pulls the rotor around. The rotor is the squirrel cage type described earlier. The reason this motor is so simple (and hence reliable) is that it is self-starting—just apply the power, and it starts.

The three-phase motor can be wired to run in either direction by simply reversing any two of the three leads. For example, if the three motor terminals are connected to phases ABC and it turns CW, then if wires B and C are reversed (to ACB), the motor will rotate CCW. (Reversing C and A then brings it back to CW.)

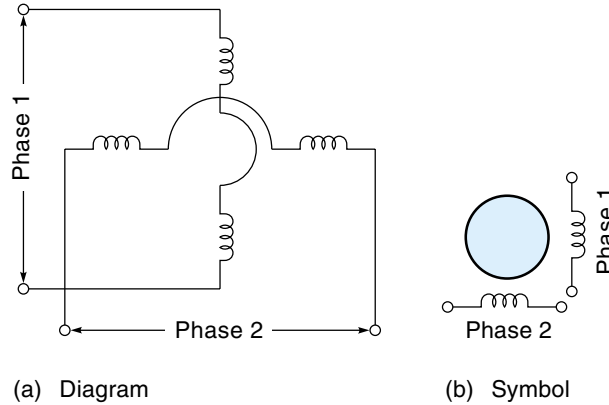
A three-phase motor, once started, will continue to run even when one of the phases is disconnected, because two-thirds of the rotating field is still working and the mechan-

**Figure 9.20**  
A three-phase induction motor.



**Figure 9.21**

A two-phase motor.



ical inertia of the spinning rotor will carry it over the “dead spot” caused by the missing wire. However, vibration and noise will increase, torque will decrease, and the motor may overheat due to greater current in the active field windings.

### Split-Phase Control Motors

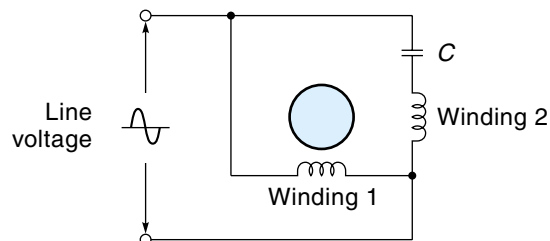
The **split-phase control motor** is technically a two-phase motor because it has two sets of windings (Figure 9.21). Not as common as the three-phase or single-phase types, these motors do find application in control systems. The operating parameters that make them desirable are (1) they are self-starting and (2) they can easily be controlled to turn in either direction. Typically, they are small (less than 1 hp) and are used to move something back-and-forth, such as opening and closing a valve or raising and lowering a garage door.

The problem is that two-phase AC is not available directly from the power company; it must be created, usually from single-phase AC. This adds some complexity to the system, but it does provide the opportunity to control the direction of rotation of the motor and hence its value as a back-and-forth type of actuator in a control system.

The required two-phase power is created from single-phase AC by placing a capacitor in series with one of the windings (Figure 9.22). The capacitor causes the current in winding 2 to lead the line current in winding 1 by almost  $90^\circ$  (it doesn't have to be exact). To change the direction of rotation, the capacitor must be able to switch so that it is in

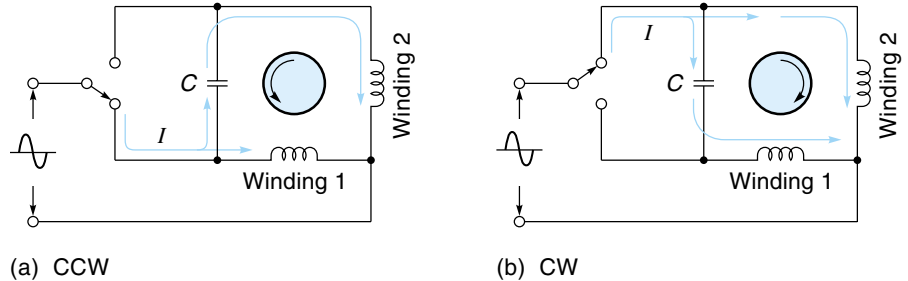
**Figure 9.22**

A split-phase motor (two-phase) driven with single-phase AC.



**Figure 9.23**

Controlling the direction of a split-phase control motor (two-phase motor).



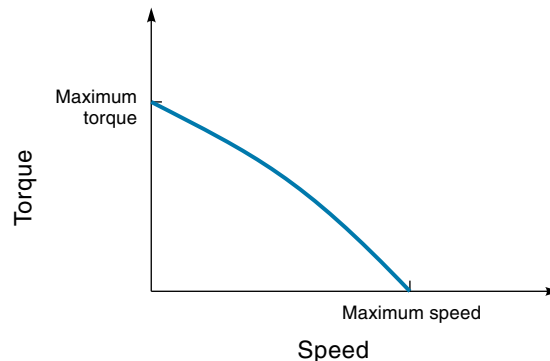
series with the other winding. Such a circuit is shown in Figure 9.23(a). Notice that the switch is in the down position, which causes the line voltage to be applied directly to winding 1, while winding 2 is fed through the capacitor. In Figure 9.23(b), the switch is in the up position, which allows winding 2 to get the line voltage, while winding 1 is fed through the capacitor. This causes the motor to rotate in the opposite direction.

### AC Servomotors

A special case of the two-phase motor is the **AC servomotor**. This is a high-slip, high-torque motor, designed specifically for control systems, and it has a relatively linear torque-speed curve (Figure 9.24). As you can see, the maximum torque occurs when the speed is zero. When the motor is running, the speed is inversely proportional to the load torque; put another way: the lighter the load, the faster the motor runs. This is very similar to the way a DC motor behaves. Figure 9.25 shows a diagram of the AC servomotor. The two windings are called the **main winding** and the **control winding**. The main winding is connected to an AC source, usually 120 Vac. The control winding is driven by an electronic circuit that (1) causes the phase to be either leading or lagging the main winding (thereby controlling the motor direction) and (2) sets the magnitude of the control-winding voltage, which determines the speed. Typically, the maximum control winding voltage is about 35 Vac. If the control winding has 0 V, the motor will coast to a stop, even though the main winding is still connected to the line voltage. This is different from a normal induction motor that *will* continue to run on a single phase.

**Figure 9.24**

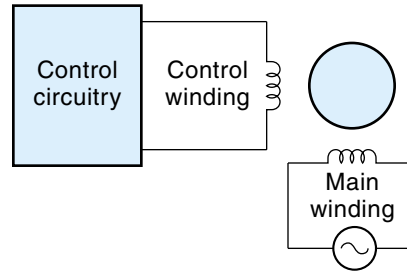
AC servomotor torque-speed curve.





**Figure 9.25**

Diagram of an AC servomotor.



### 9.3 SYNCHRONOUS MOTORS

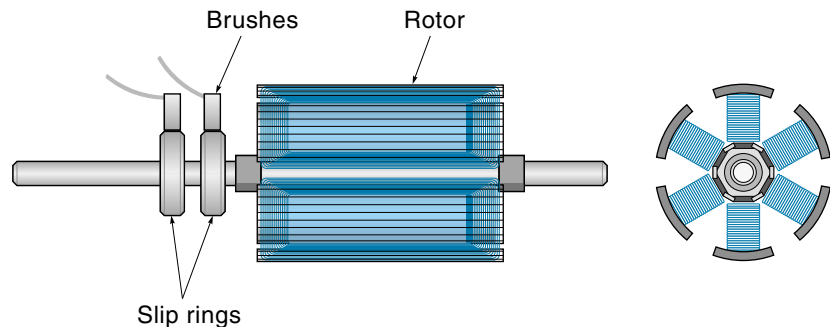
The **synchronous motor** is similar to the induction motor with one important difference: The rotor in the synchronous motor rotates at exactly the speed of the rotating field—there is no slip. Put another way, the speed of the synchronous motor is always an exact multiple of the line frequency. This feature is extremely desirable in industrial applications, for example, when several motors along a conveyer belt must all be going exactly the same speed. Although many synchronous motors are large, the concept is also used extensively in small clock or timing motors where an exact relationship must exist between frequency and speed.

#### Theory of Operation

Recall that the rotor in an induction motor receives its power through induction, which requires a difference (slip) between the speed of the rotor and the rotating field. To make a synchronous motor work, the power to form a magnetic field in the rotor must come from another source. Traditionally, this is done by supplying DC power into the rotor via **slip rings** and brushes (Figure 9.26). Slip rings and brushes on the synchronous motor are similar to the commutator assembly used in DC motors, with one important difference; here the electrical contact from stator to rotor is made through a smooth ring, not the multiple contacts of the DC motor's commutator. The action is smoother,

**Figure 9.26**

The rotor of a synchronous motor showing slip rings.



the components last far longer, and less electrical noise is generated. The rotor of the synchronous motor uses DC power (called *excitation*) to energize electromagnets around its perimeter. These magnets tend to lock on to the rotating magnetic field in the stator and cause the rotor to rotate at the exact speed of the rotating field, that is, the synchronous speed.

Unfortunately, the synchronous motor is not self-starting. When the power is switched on, the field starts rotating full speed, causing the magnetized rotor to be first tugged in one direction and then the other. Experiencing no sustained torque, the rotor just sits there. To start the motor, additional hardware is required. Any one of the following methods can be used:

1. Provide for the frequency to start slow and then speed up. As long as the rotating field does not get too far ahead of the rotor, the rotor will continue to be tugged in one direction. The problem is that it takes extra electronics to provide the variable frequency. However, the advantage of this system is that if you can control the frequency, you can control the speed. This approach is being used more and more, with the continuing advancement of high-power solid-state electronics.
2. Use another motor (called a **pony motor**) to bring the synchronous motor up to speed. This approach has the obvious drawback of requiring significant additional hardware.
3. Insert some squirrel cage rotor bars (as used in induction motors) in the rotor of the synchronous motor. This allows the motor to get started as if it were an induction motor.
4. Use hysteresis-synchronous motors (covered later in this chapter).

## Power-Factor Correction and Synchronous Motors

The AC power from the power company arrives with the voltage and current in phase with each other. This is the most efficient way to transmit power because, by definition, power is the product of voltage times current. Therefore, for maximum power transfer, the voltage and current must be occurring at the same time, that is, in phase.

The windings of induction motors present an inductive-type load to the power line; if there are many motors in a facility, this inductive load can be significant. The problem is that the inductive load causes the current to lag behind the voltage, which means the motor will draw more current to deliver the same power. This situation is described in terms of the power factor. The **power factor** is the cosine of the angle between the voltage and current, so when the voltage and current are exactly in phase, the power factor is 1. For a purely reactive load of a  $90^\circ$  phase shift between voltage and current, the power factor is 0. We say that an inductive load creates a *lagging power factor*, and a capacitive load creates a *leading power factor*.

Plant engineers would like to keep the power factor as close to 1 as possible, but how can this be done with a plant full of inductive motors? First, it helps to have the

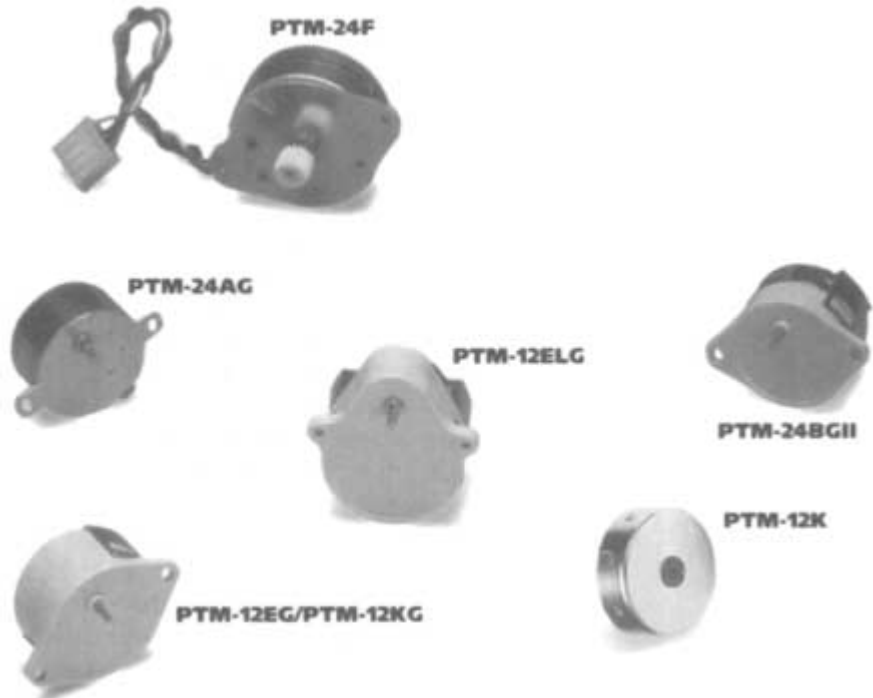
induction motors running with a full load—a lightly loaded induction motor is much more inductive than a fully loaded one. Beyond that, the way to improve a lagging power factor is to add a capacitive load to cancel out all or part of the inductive load. There are two ways to implement this; one way is to connect large power-factor-corrective capacitors across the lines, and the second way (surprisingly) is to make use of synchronous motors. With proper adjustment of the rotor excitation current, the synchronous motor will present a capacitive load, which will improve (increase) the power factor of the plant. Synchronous motors installed specifically for power-factor correction are called **synchronous condensers**; they do not drive a load—they simply spin.

### Small Synchronous Motors

Small synchronous motors are used as timing or clock motors or simply as a source of rotary power in instruments and small machines. They are available to be used as a direct drive or with an attached gear train. Figure 9.27 shows a selection of these motors. Many of these motors are split-phase motors designed to operate on single-phase AC

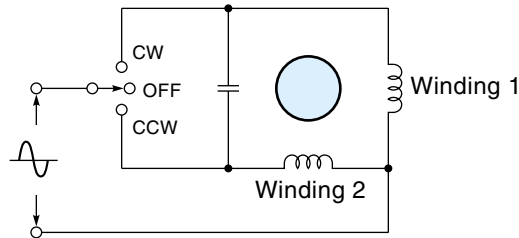
**Figure 9.27**

A selection of small synchronous motors. (Courtesy of Kollmorgen Motion Technologies Group)



**Figure 9.28**

A small synchronous motor being driven with single-phase AC (switch controls direction of rotation).



with the use of a capacitor. The simple switching arrangement shown in Figure 9.28 allows for the motor to rotate in either direction. When the switch is in the up position, winding 1 gets the line voltage and winding 2 is phase-shifted through the capacitor. This arrangement causes the motor to turn in one direction. When the switch is in the middle position, the motor is off; when the switch is in the down position, winding 2 gets the line voltage, and winding 1 is driven through the capacitor. This causes the motor to turn in the other direction.

Small synchronous motors do not use electromagnets in their rotors and so do not need slip rings. Instead, the rotor is magnetized in other ways. There are three designs: the reluctance motor, the hysteresis motor, and the permanent magnet motor. All three types can be made to be self-starting. The hysteresis motor is particularly interesting because the rotor consists of a smooth, round piece of steel. Its operation is very smooth and quiet, which makes it an ideal choice for such applications as tape drives.

## 9.4 UNIVERSAL MOTORS

The **universal motor** is so named because it can be powered with either AC or DC. Basically, it is a series-wound DC motor that has been specifically designed to operate on AC. Like its DC counterpart, it is reversible by changing the polarity of either the field or the rotor windings, but not both. Physically, the universal motor is similar to a DC motor except that more attention is paid to using laminations (thin sheets of lacquered metal) for the metal parts (to reduce the AC eddy currents) and the inductance of the windings is minimized as much as possible.

The operating characteristics of the universal motor are similar to those of the DC motor. For a fixed voltage, the speed is inversely proportional to the load torque—as the load increases, the speed decreases. For a constant load, as the applied voltage increases, the speed will increase. Typically, universal motors are designed to operate at high speeds—from 3600 to 20,000 rpm—but, because they use a commutator and brushes (which wear out), they have a limited lifetime. Being a series-wound motor, they have high starting torque, and for this reason are widely used for handheld power tools (for example, a hand drill motor).

## 9.5 AC MOTOR CONTROL

When the AC motor is first switched on, there is an initial rush of current that may be five times higher than the average running current. In fractional hp motors, this momentary high current is usually acceptable, but for larger motors special provision is usually made to reduce the initial current surge to prevent electrical and mechanical components from being overstrained. Once the motor is running, its speed will be determined by the frequency, which is why the AC motor is usually thought of as a constant-speed motor. At one time, “controlling” an AC motor meant “turning it on and off.” Currently, this notion of the AC motor being constant speed is being revised because efficient electronic power circuits can vary the frequency, thus allowing the speed to change. This and other topics having to do with motor control are the subjects of the subsequent sections.

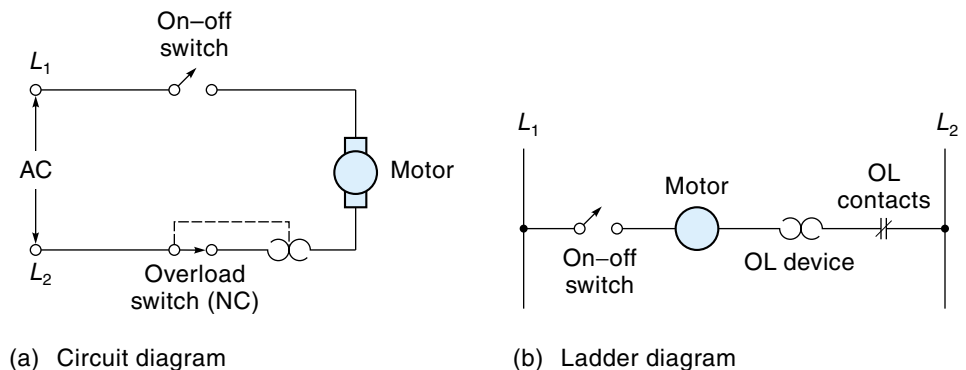
### Start–Stop Control

The simplest way to start a fractional-hp AC motor is to connect it to the line voltage through a switch, and this is how it’s done in many cases. Figure 9.29 shows two different schematics for this approach. Both schematics are for the same circuit, but Figure 9.29(b) is drawn in a form called a **ladder diagram** and is typical of an industrial motor-wiring diagram. In the ladder diagram, Figure 9.29(b),  $L_1$  and  $L_2$  are the AC line voltage wires, and each “rung” is a circuit. (Ladder diagrams are covered in Chapter 12.)

The symbol labeled “OL” in Figure 9.29(b) represents the normally closed contacts of the **overload device** (normally open contacts would not have the diagonal line). The overload device is designed to open its contacts if the motor gets too hot. This device may be a thermal sensor built into the motor that will trigger when a certain temperature is reached, or it may be separate from the motor—in which case it is actually triggering on a sustained current-overload condition.

**Figure 9.29**

A simple on–off motor-control circuit.



The start-stop circuit for motors used in industrial processes is typically a little more complicated than an on-off toggle switch. The motor might well be controlled from a control panel some distance from the motor. In such a case, routing the motor power through a relay located near the motor and running the low-current wires (which only need to energize the relay coil) to the control panel make sense. This approach is particularly appropriate for larger motors. An everyday example of this system is the starter motor in a car. When you turn the key, a low-current ignition switch provides power to energize a high-current capacity relay (starter solenoid) that, in turn, connects the starter motor to the battery.

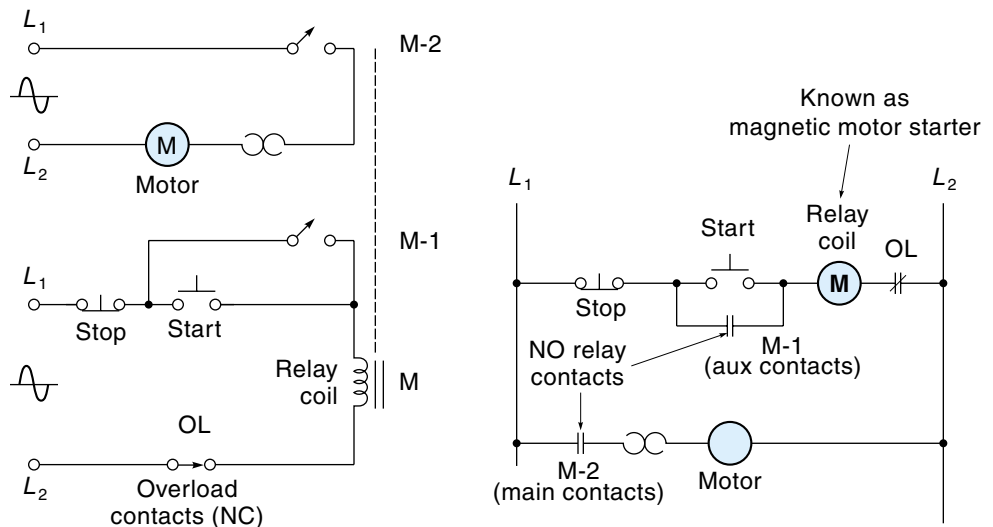
A relay specially built to start motors is known as a **magnetic motor starter** or **magnetic motor contactor** (see Figure 9.32). A magnetic motor starter has large contacts to handle the motor current and often includes a built-in current overload device. When the current exceeds some pre-set value, the contacts open.

Another problem with using a simple toggle switch for motor control has to do with safety. Imagine the following situation: A motor driving a potentially hazardous machine overheats, and the overload protection circuit shuts it down. Somebody comes along, sees the motor is stopped, and investigates. Meanwhile, the motor cools down, the thermal protection device resets, and—because the power switch is still on—the motor suddenly restarts. The unsuspecting investigator may get hurt.

The standard industrial motor-control circuit has been designed to address both of these situations. The motor is connected to the power lines through a magnetic motor starter (relay) and is controlled with separate start and stop buttons on a control panel. To put it simply, the motor is started by pushing the start button and stopped by pushing the stop button. If the motor stops because of overheating, the start button *must* be pushed again to restart it. Figure 9.30(a) shows the circuit diagram for the start-stop circuit. The

**Figure 9.30**

A start-stop circuit for a single-phase AC motor.



stop button is normally closed, and the start button is normally open. When the start button is pushed, power is applied to motor starter relay coil M, which closes contacts M-1 and M-2. Contact M-2 connects the motor to the AC power, and the motor starts. Contact M-1 connects the relay coil to a point in the circuit ahead of the start button; consequently, when the start button is released, the relay coil will remain energized. This process is called **sealing**, or **latching**, the relay. When the stop button is pushed, power is removed from the starter relay coil, which opens contacts M-1 and M-2. The opening of the M-2 contact stops the motor, and the opening of the M-1 contact “breaks the seal” that was keeping the relay coil energized. Consequently, when the stop button is released, the relay coil remains unenergized, and the motor stays off. If the overload contact should open at any time, the result is the same as when the stop button is pushed: The motor stops and will not restart until the start button is pushed. A ladder diagram of the start-stop circuit is shown in Figure 9.30(b). Notice that the relay coil is represented by a circle.

Besides safety reasons, sealing circuits are useful when the entire plant loses power. When the power is restored, you don’t want all motors starting up at the same time, causing a tremendous current surge in the main power lines. Instead, the motors can be restarted individually or in groups, in some orderly fashion. Also, starting and stopping motors from multiple locations is common; for example, a cooling pump in a nuclear power plant might be started from three places: for example, the main control room, the pump site, and an auxiliary control room.

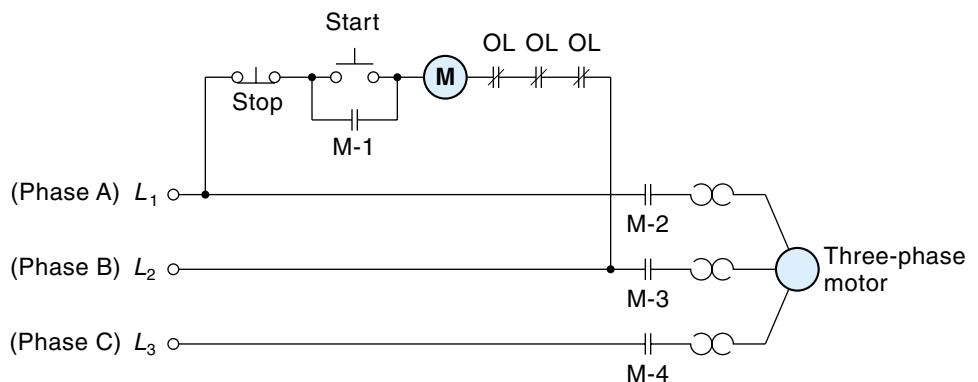
Figure 9.31 shows the start-stop circuit for a three-phase motor. This is basically the same as for the single-phase motor except that the relay now provides three sets of contacts for the motor (M-2, M-3, and M-4). Also, the motor starter coil circuit, which requires only single-phase AC, is connected across  $L_1$  and  $L_2$ . For larger motors, the coil circuit voltage may be stepped down to 120 V with a transformer.

## Jogging

**Jogging** refers to applying power to the motor in short bursts for the purpose of “inching” it into some desired position. For example, a commercial mixer might need to be

**Figure 9.31**

A start-stop circuit for a three-phase AC motor.



**Figure 9.32**

A magnetic motor starter. (Allen-Bradley products courtesy of Rockwell Automation.)

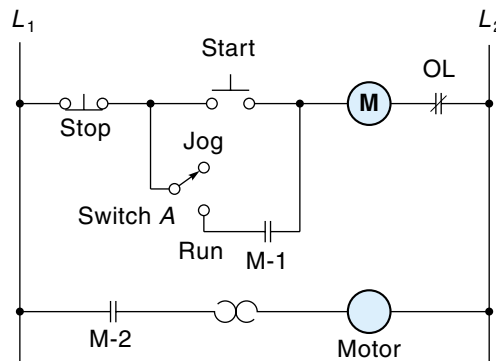


jogged into a certain position so the blades can be replaced, or a motor driving a large shaft might need to be jogged to get the keyway lined up with the mating part.

The traditional motor-control circuit discussed in the previous section (which uses a sealing relay) does not lend itself to jogging because it's not responsive enough—that is, it can't cycle on and off fast enough. Jogging requires that you have direct push-button-power control to the motor, which usually means bypassing the regular on-off relay circuit. Figure 9.33 shows a simple motor-control circuit that has provision for jogging. When switch A is set to start, this circuit behaves exactly like the circuit of Figure 9.30;

**Figure 9.33**

A start-stop circuit with a provision for jogging.





that is, the relay is sealed every time the start button is pushed. However, when switch A is set to jog, the sealing circuit is disabled, and power is supplied to the motor only when the start button is pushed.

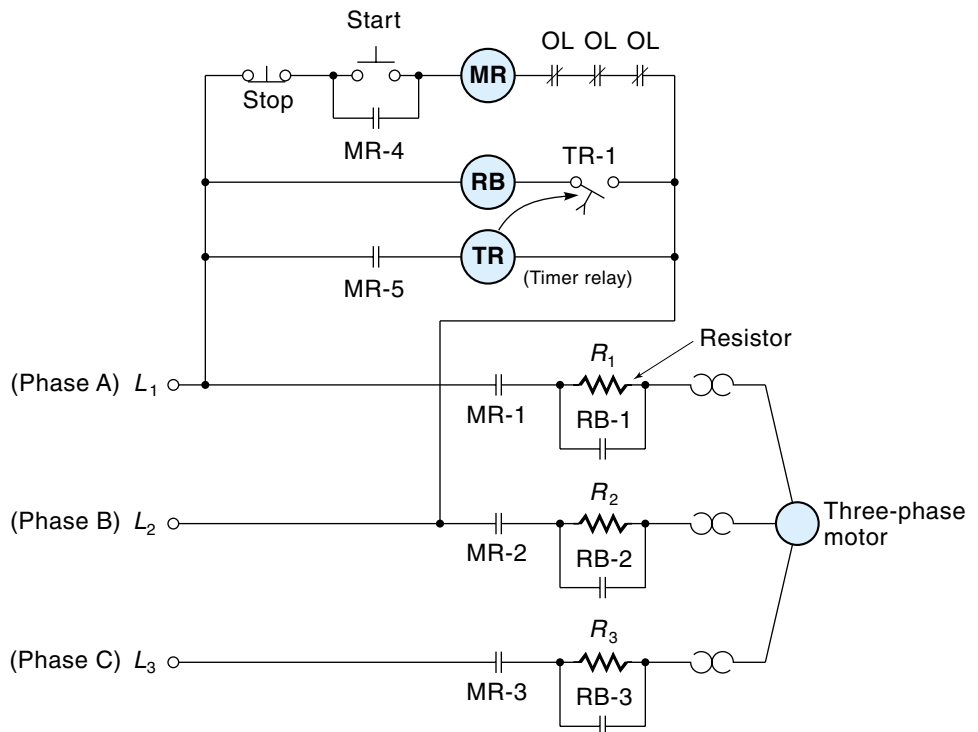
### Reduced-Voltage Starting

The nature of any motor is to have a large initial current when first connected to a voltage source. The power-distribution system can usually cope with the increased start-up current of fractional hp motors (although the lights may dim perceptibly in the immediate area), but larger motors usually need special starting circuits that limit the starting current to some acceptable value. One of the simplest ways to limit start-up current is to connect resistors in series with the motor during the start-up period. After the motor has come up to speed and is drawing a reduced amount of current, a switching device removes the resistors from the circuit, and the motor is connected to the full line voltage.

Figure 9.34 shows a reduced-voltage starting circuit for a three-phase motor. Notice the presence of resistors  $R_1$ ,  $R_2$ , and  $R_3$  in series with the motor power lines. Also notice that each resistor is bypassed with a relay contact. When the start button is pushed, the MR relay coil is energized, and contacts MR-1, MR-2, and MR-3 close. This action

**Figure 9.34**

A reduced-voltage starting circuit.



connects the motor to the power lines, but the current must go through the resistors. Another MR contact (MR-5) applies power to the coil of a **timer relay** (TR), a relay with a built-in delay mechanism (based on pneumatics or electronics) that causes its contacts to close a certain time *after* the TR coil is energized. The delay is set to give the motor enough time to come up to near-operating speed. When the relay “times out,” contact TR-1 closes and allows relay coil RB to be energized. Relay RB has three sets of contacts (RB-1, RB-2, and RB-3) that connect the motor directly to the power lines, bypassing the resistors.

The reduced-voltage starting circuit is simple and relatively inexpensive, but it does have the drawback of reducing the motor start-up torque. Remember that torque is proportional to current, and if current is limited, so is the torque. One way to improve the situation is to use transformers instead of resistors in the start-up circuit. The advantage of using transformers (actually autotransformers) is that they can provide the motor increased current, while not straining the power-distribution system. The transformers are only engaged during the start-up period. When the motor gets up to speed, the transformers are switched out, and the motor is powered directly from the power lines.

## Variable-Speed Control of AC Motors

AC motors enjoy many advantages over DC motors: They are lighter in weight, more reliable, less expensive, and require less routine maintenance. And although small speed changes can be made by changing the voltage, AC motors are essentially limited to running at a speed determined by the line frequency. This fact has traditionally kept them from being used in control systems where the speed needs to be varied. The introduction of high-power solid-state switching circuits has changed this picture. Clearly, if you want to fully control the speed of an AC motor, you must be able to change the frequency. This can be done with off-the-shelf power-conversion units that are capable of converting the line voltage at 60 Hz into a wide range of voltages and frequencies. A motor-control unit (or the control unit plus the motor) is called a **drive**. There are four choices of AC-motor variable-speed drives: the older variable-frequency drive (also known as a V/Hz drive) and the newer vector drives (sensorless, flux vector, and field-oriented control).

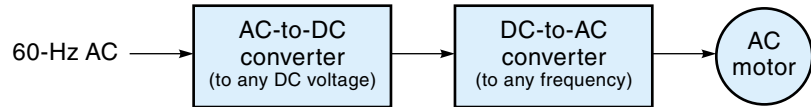
## Variable-Frequency (V/Hz) Drives

The traditional **variable-frequency drive** (known as a **volts-per-hertz (V/Hz) drive**) changes the motor’s frequency and voltage using solid-state control units.

The basic steps for this process are shown in the block diagram of Figure 9.35, and the circuit is known as a **DC link converter**. The first step is to convert 60-Hz AC into DC power. The second step is to convert this DC power back into AC at the

**Figure 9.35**

A block diagram of a variable-speed-control system.



desired frequency. Figure 9.36 shows a typical generalized circuit to perform these functions for a three-phase motor. In this circuit, 60-Hz AC line power is converted to DC with a silicon-controlled rectifier (SCR) network. SCRs are used so the magnitude of the DC voltage can be controlled. Recall that the SCR is essentially a diode, but in order for it to conduct, two conditions must be met. First, it must be forward-biased; second, it must receive a trigger pulse on its gate input. Referring to the circuit of Figure 9.36, if the gate (G) of each positive AC cycle, the conduction time will be maximum, and the maximum DC voltage will result. If the trigger pulse to the gate is delayed, then the SCR is conducting for a shorter time, and a lower DC voltage is established (note that the  $L$  and  $C$  in Figure 9.36 help smooth out the power pulses from the SCRs). To sum up, the average value of the DC voltage can be determined by controlling the delay of the trigger pulses.

The next job of this circuit (Figure 9.36) is to create a sort of artificial three-phase AC power at any desired frequency. This is accomplished with the six transistors on the right side of the circuit. Each transistor is turned on and off in sequence by a controller circuit (not shown) in such a way as to cause three pseudo-sine waves.

**Figure 9.36**

A variable-speed-control circuit (trigger and timing circuits not shown).

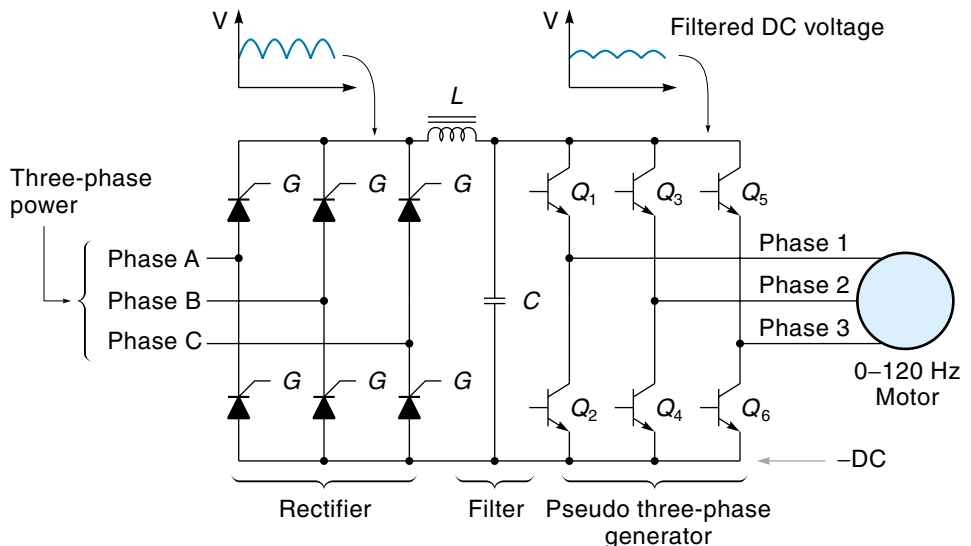
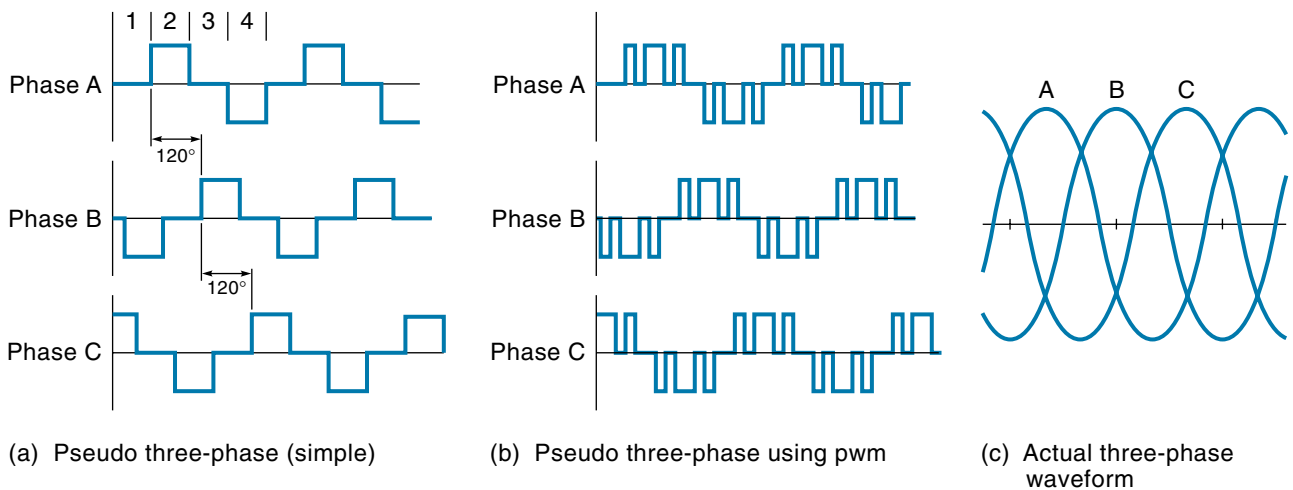


Figure 9.37(a) shows the simplest three-phase output voltage waveforms that can be made from this circuit. In particular, notice the phase A voltage waveform. During time period 1, both transistors ( $Q_1$  and  $Q_2$ ) are off, so the output, which is taken from between the transistors, is neither positive or negative. Then during time period 2, transistor  $Q_1$  is on, connecting the phase A output to the plus DC voltage. During time period 3, both transistors are again off; finally, during time period 4, transistor  $Q_2$  is on, connecting the phase A output to the minus side of the power supply. This same-shape waveform is generated by transistor pair  $Q_3$ - $Q_4$  and again by  $Q_5$ - $Q_6$ , with each phase lagging the one ahead of it by  $120^\circ$ . Clearly, the apparent frequency of the output is determined by how fast transistors  $Q_1$ - $Q_6$  are sequenced, (which is typically in the range of 5–120 Hz). To help visualize the three-phase action, compare the phases A, B, and C in Figure 9.37(a) with phases A, B, and C in the traditional three-phase diagram [Figure 9.37(c)]. A better pseudo-AC waveform can be created by using PWM (pulse width modulation) as shown in Figure 9.37(b).

For the motor to work well at various speeds, the voltage to the motor must be modified each time the frequency is changed. Specifically, the voltage and frequency should be held proportional—that is, when the frequency is increased, the voltage should be increased, and vice versa. The reason for this requirement is that the current in the stator windings must be maintained at a certain design value for the magnetic induction process (to the rotor) to work. Most motors are designed to operate at 60 Hz and 120 V (or 240 V), so the stator is wound to create the proper magnetic field with those conditions. If the frequency drops below 60 Hz, the inductive impedance of the windings also drops, which would allow in more current. Consequently, the voltage must be lowered as the frequency is lowered in order to maintain the proper stator current. Figure 9.38(a) shows how the voltage should increase linearly with frequency in the range 0–60 Hz and explains why this type of drive is called a *volts-per-hertz* or V/Hz type. The voltage is usually not allowed to increase beyond the motor's rated voltage (for its own health). So,

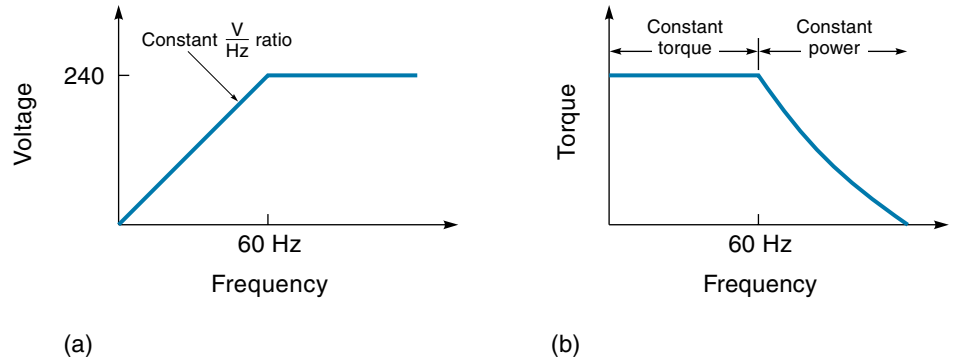
**Figure 9.37**

Output-voltage waveforms of a speed-control circuit.



**Figure 9.38**

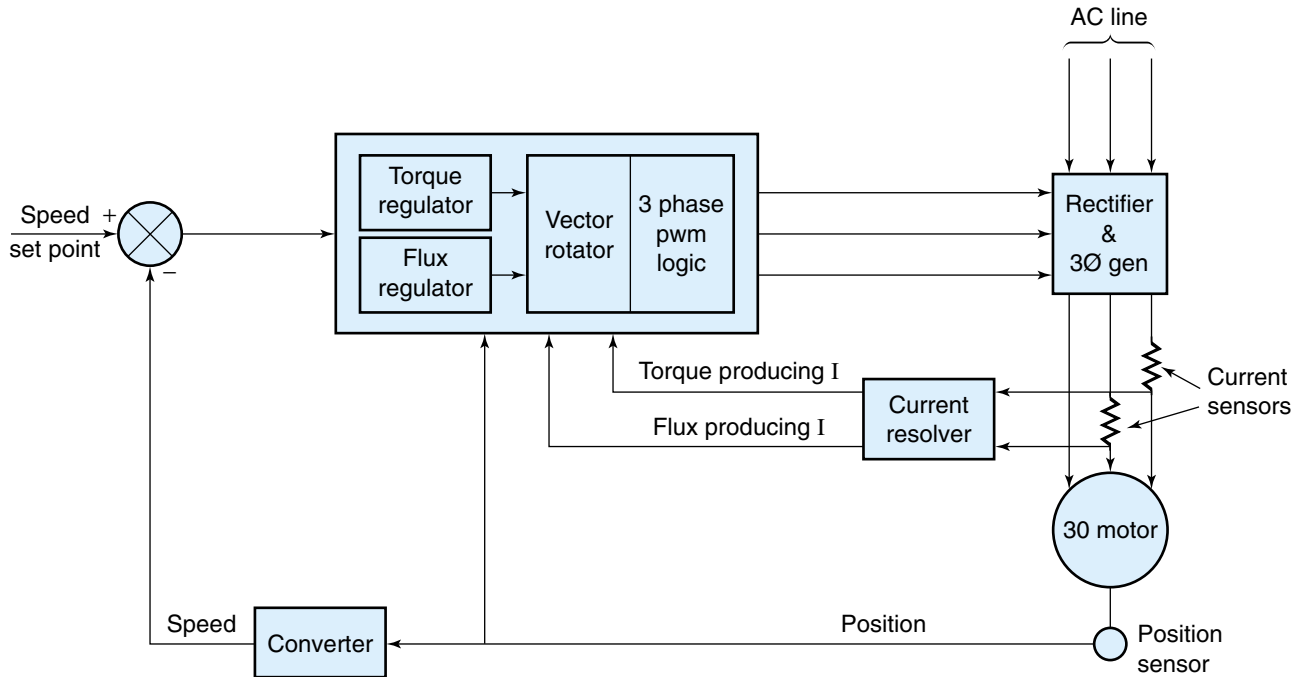
Voltage and frequency are controlled in a variable-speed-control circuit.



in practice, there are really two distinct operating ranges. The first range (5-60 Hz) is called the constant-torque region because the motor produces a constant torque in this speed range, as shown in Figure 9.38(b). This is the same torque that the motor has at normal (60 Hz) operating speed. The region above 60 Hz is known as the *constant-power region* because, even though the torque is falling off, the speed is increasing, so the actual mechanical power stays the same (power is the product of speed times torque). Commercial motor-control circuits are capable of meeting these qualifications.

## Vector Drives

**Vector drives** are based on the principle that the current driving an AC induction motor can be divided into two components: the current that produces the magnetic field flux in the stator and the current that creates the torque that causes rotation. The actual motor current is the *vector* sum of these two currents, and if they can be independently controlled, it is possible to drive the motor at full torque at any speed, right down to 0 Hz. There are three types of vector drives: sensorless vector, flux vector, and field-oriented control drives. We will discuss the **field-oriented control drive** first, because it is the newest and provides the most control. A block diagram is shown in Figure 9.39. First, you should realize that the motor can be any standard induction motor, but the drive is a very sophisticated microprocessor-driven feedback-control system. As you can see, the motor itself is driven by variable-frequency circuit similar to that shown in Figure 9.36. Current sensors on the motor leads feed a *current resolver*, which identifies the flux-producing and torque-producing currents in the motor. A position sensor mounted on the shaft provides position and speed information. The drive electronics uses all this information to maintain two independent control loops: a speed/torque loop for control of the motor speed and torque, and a flux loop to provide a constant magnetizing stator current throughout the motor's speed range. By maintaining a constant stator flux, the motor is capable of providing a constant torque from its base frequency (60 Hz) all the way down to 0 Hz. The speed/torque loop is able to determine what the moment-to-moment slip is (some slip is necessary for any induction motor) and then compensate by adjusting the



**Figure 9.39**  
Block diagram  
of a vector AC drive  
(field-orientated type).

rotational speed of the field very precisely within the windings so as to make up for the slip. This process can keep the motor running at exactly its set-point speed, regardless of the torque demand. The outputs of both the flux- and the speed/torque-control loops are combined in the *vector rotator* to produce a single set of three-phase voltage waveforms. These waveforms are converted into PWM and fed to the transistors in the variable-frequency generator.

The **sensorless drive** predates the field-oriented control drive and cannot provide such precise control. As the name suggests, it does not require a position sensor but instead makes “guesses” based on current feedback and what it knows about the motor. This system is adequate for many applications.

The **flux vector drive** does require a position sensor and maintains better control than the sensorless drive can (but not as good control as the field-oriented drive). This system estimates the flux-producing and torque-producing current vectors in the motor and uses this information to control the motor. Flux vector drives can function effectively down to 2–3 Hz.

## SUMMARY

The AC motor is a common source of electric rotary power. Most AC motors do not use commutator/brushes, which makes them more reliable and less expensive than DC

motors. The primary use of AC motors is for constant-speed applications because the speed of an AC motor is dependent on the line frequency.

AC power is supplied in two forms: single phase and three phase. Single-phase AC is the common form supplied to homes and businesses and requires two wires. Three-phase power is usually only available in industrial locations and consists of three separate 60-Hz sine waves,  $120^\circ$  apart, on three wires (four including the neutral).

The most common type of AC motor is the induction motor. In the induction motor, power is supplied only to the stator windings, where it creates a rotating magnetic field around the rotor. The rotor receives some electric power through induction from the stator, which it uses to generate a magnetic field. This field is attracted to the stator's rotating field, which causes the rotor to rotate (but slightly slower than the rotating field).

Motors are designed to run on either three-phase power or single-phase power. Three-phase motors are the simplest because they are self-starting, but they can only be used where three-phase power is available. Single-phase motors are not inherently self-starting. There are a number of ways used to start single-phase motors, and most involve adding a second set of stator windings (called a split-phase motor), which are only energized during start-up.

The synchronous motor is designed to rotate at an exact multiple of the line frequency. For larger motors, this means that the rotor receives its power through slip rings and brushes, which complicate the design (induction will not work in this application). Smaller synchronous motors (timing motors) do not use slip rings.

The universal motor is so named because it can run on AC or DC. It is essentially a DC motor (with brushes) designed to minimize the losses that occur when running on AC.

AC motors have a large initial current when first switched on. For larger motors, a special starting circuit may be required to prevent damage to the power system and/or motor. It works by reducing the voltage to the motor during start-up, either by resistors or transformers.

The speed of an AC motor can be controlled in two ways. A small amount of speed control can be had from simply reducing the voltage, which causes the rotor speed to slip more. For complete speed control, the frequency must be controlled with control circuits that convert 60-Hz AC to other frequencies. The simplest of these systems is called a variable-frequency (or K/Hz) drive. Better motor performance is possible with newer, more sophisticated units called vector drives.

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## GLOSSARY

**AC servomotor** A two-phase motor used in control applications; designed to have a near-linear torque-speed curve.

**capacitor start motor** A single-phase AC induction motor that uses a capacitor to phase-shift the AC for the start windings.

**control winding** One of two windings in an AC servomotor.

**DC link converter** A circuit that converts line voltage at 60 Hz to AC power at different voltage and frequencies, for the purpose of controlling the speed of AC motors.

**delta-connection** One of two ways in which to connect three-phase motor or generator coils. With the delta-connection, each of the three coils is connected between two phase wires; there is no neutral.

**field-oriented control drive** A class of AC-motor-control units that provides very precise control of speed. A type of vector drive. See **vector drives**.

**flux vector drive** A class of AC-motor-control units that requires a position sensor and maintains better control than the sensorless drive can (but not as good control as the field-oriented drive). A type of vector drive. See **vector drives**.

**fractional-horsepower (hp) motor** A motor with less than 1 hp.

**frame** A basic part of an AC motor; an overall case that supports the stator and the rotor bearings.

**GFI** See **ground-fault interrupter**.

**ground-fault interrupter (GFI)** A safety device that will shut off the power if it senses a current leakage to the safety ground.

**integral-horsepower (hp) motor** A motor with 1 hp or more, that is, a large motor.

**jogging** The practice of “inching” a motor into position by repeated short bursts of power.

**ladder diagram** A type of wiring diagram typically used for industrial motor-control circuits.

**latching** See **sealing**.

**line voltage** A single number that represents the voltage of a power system; for three-phase power, the line voltage is the voltage between any two lines.

**magnetic motor contactor** A special-purpose relay used to switch power to a motor. The magnetic motor contactor may have large contacts to handle the motor current and may have a built-in current overload trip device.

**magnetic motor starter** See **magnetic motor contactor**.

**main winding** One of two windings in an AC servomotor.

**neutral** The return wire or cold side in an AC power-distribution system; the neutral wire is at ground potential.

**overload device** A device that can stop the motor if an overload condition is detected. Overload is detected by sensing the temperature of the motor or the current that the motor is drawing.



**permanent-split capacitor motor** A capacitor start motor where the start winding remains engaged during operation—that is, the motor does not have a centrifugal switch.

**phase sequence** The sequence of phase voltages possible with three-phase power: ABC and ACB.

**phase voltage** The individual load voltage in a multiphase power system.

**pony motor** A separate motor used to start a synchronous motor.

**power factor** The cosine of the angle between the current and voltage. A power factor of 1 means the current and voltage are in phase, which is desirable. A lagging power factor means the current is lagging the voltage, probably due to the inductive load of motors.

**pull-out torque** The maximum torque that an AC motor can provide just before it stalls. However, unlike the DC motor, this maximum torque condition occurs at about 75% of the unloaded speed.

**return wire** *See neutral.*

**rotating field** The phenomenon of what is apparently happening in the stator coils of an AC motor. Even though the coils themselves are stationary, the magnetic field is transferred from coil to coil sequentially, producing the effect that the field is rotating.

**rotor** A basic part of an AC motor, the part in the center that actually rotates.

**run windings** The main stator windings in a single-phase motor.

**sealing** A process whereby a relay is initially energized by a start button and then stays energized from current coming through one of its own closed contacts.

**sensorless drive** A class of AC-motor-control units that does not use a sensor. A type of vector drive. *See vector drives.*

**shaded-pole motor** A single-phase induction motor, usually small, that does not use a start winding. Instead, a modification to the single-phase poles causes the motor to have a small starting torque.

**single-phase AC** The type of AC supplied to residences (and industry). It consists of a single alternating waveform that makes 60 complete cycles per second.

**slip** The difference between the rotor speed of an induction motor and the synchronous speed of the stator field; the amount of slip is typically less than 10%.

**slip rings** Sliding electrical contacts that allow power to be fed to the rotor of the synchronous motor.

**speed regulation** The ability of a motor to maintain its speed under different loads.

**split-phase** A general term applied to an induction motor that has two sets of windings but operates on single-phase AC with some provision to phase-shift the AC for the second set of windings.

**split-phase control motor** A two-phase motor, usually small, powered by single-phase AC using a capacitor for the phase shift. The direction is reversible, making these motors useful in controlling back-and-forth motion.

**squirrel cage rotor** The type of rotor used in an induction motor—named as such because someone thought it looked like a squirrel cage.

**stall** When the load on the AC motor is increased (about 75% of unloaded speed), the motor torque drops back abruptly, and the motor stalls (stops).

**starting torque** The motor torque available when the motor is first started; the starting torque of most AC motors is less than its maximum torque but more than its rated load.

**start windings** A second set of windings (besides the run windings) in the single-phase induction motor; the start windings are usually only engaged during the starting process.

**stator** A basic part of an AC motor, a collection of coils that surrounds the rotor and does not move.

**synchronous condenser** A term applied to a synchronous motor that is being used solely for power-factor correction.

**synchronous motor** An AC motor that runs at the synchronous speed, which means an exact multiple of the line frequency. A synchronous motor has no slip.

**synchronous speed** The speed of the rotating field in the stator, which is always an exact multiple of the line frequency. A synchronous motor spins at the synchronous speed, whereas an induction motor turns somewhat slower than the synchronous speed.

**three-phase AC** The type of AC available to industry. It consists of three AC waveforms (called phases) on three wires, where each phase is delayed 120° from the previous phase and all phases operate at 60 Hz.

**timer relay (TR)** A relay with a built-in time delay mechanism so that the contacts close or open a specified time after the relay is energized.

**two-capacitor motor** A capacitor start motor that uses one value of capacitance for starting and switches to a lower value of capacitance for running.

**TR** See **timer relay**.

**universal motor** An electric motor that can run on AC or DC power; it is essentially a series DC motor.

**variable-frequency drive** (known as a **volts-per-hertz (V/Hz) drive**) A type of AC-motor-control unit that controls speed by changing the motor's frequency and voltage using solid-state control units.

**vector drives** A class of AC-motor-control units that are based on the principle that the motor current is the vector sum of the flux-producing current and the speed/torque-producing current. Some vector drives can control an AC motor down to 0 rpm.

**volts per hertz (V/Hz) drive** See **variable-frequency drive**.

**wye-connection** One of two ways in which to connect three-phase motor or generator coils; with the wye-connection, each of the three coils is connected between a phase wire and neutral.

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## EXERCISES

### Section 9.1

1. A single-phase voltage waveform is shown in Figure 9.2.
  - a. What is the voltage between prong A and ground at  $270^\circ$ ?
  - b. What is the voltage between prong B and ground at  $90^\circ$ ?
  - c. What is the voltage between prong A and prong B at  $360^\circ$ ?
2. Explain the distinguishing features of the AC waveform shown in Figure 9.5.
3. Redraw Figure 9.4 with a delta-connected load.
4. In a balanced three-phase wye-connected system, what percentage of the current flows in the neutral wire? Explain.
5. A three-phase wye-connected system has individual phase voltages of 240 Vac. Find the three-phase line voltage.

### Section 9.2

6. Explain what a *rotating field* in an AC motor is and briefly how it is created.
7. Referring to Figure 9.10, draw a diagram showing how the four coils would be energized at  $45^\circ$  (indicate the magnetic orientation).
8. Define the term *slip*. Why is slip essential for an induction motor to work?
9. An induction motor is powered by 120 Vac at 60 Hz and runs at approximately 1200 rpm. What pole motor must this be?
10. What is the synchronous speed of a 60-Hz, six-pole, three-phase motor? How many individual poles does it have?
11. What is the synchronous speed of a 400-Hz eight-pole, single-phase motor? How many individual poles does it have?
12. A single-phase 60-Hz AC motor has a rated speed of 1695 rpm. Find the slip.
13. A certain six-pole 60-Hz AC motor has a slip of 5% (under normal-load conditions). How fast is it rotating?
14. Explain the important parts of the torque-speed curve shown in Figure 9.13. Include in your explanation the following terms: synchronous speed, operating range, pull-out torque, and starting torque.

15. Is a three-phase motor reversible, and if so, how?
16. Draw a diagram of a capacitor start motor (which incorporates a centrifugal switch). Explain what happens during the start-up period. (What is the purpose of the capacitor?)
17. Draw a diagram of a permanent-split phase motor and explain how it operates.
18. Explain how a shaded-pole motor operates.
19. Draw a diagram and explain the operation of a split-phase control motor being powered by single-phase AC that can go in either direction.
20. What are the basic characteristics of an AC servomotor?

### Section 9.3

21. List the differences between the *synchronous motor* and the *induction motor*.
22. What is the purpose of the slip rings used in synchronous motors?
23. List three methods used to start a synchronous motor.

### Section 9.5

24. Describe the operation of the start-stop circuit of Figure 9.30.
25. Redraw Figure 9.31 and then include the additional circuitry necessary to have the motor start and stop in the reverse direction.
26. In the reduced-voltage-starting circuit of Figure 9.34,
  - a. What is the purpose of relay MR?
  - b. What is the purpose of relay RB?
  - c. What is the purpose of relay TR ?
27. Describe the two methods used to vary the speed of an AC motor.
28. Explain the operation of the variable-speed-control circuit shown in Figure 9.36.

# CHAPTER 10

## Actuators: Electric, Hydraulic, and Pneumatic

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### OBJECTIVES

After studying this chapter, you should be able to:

- Understand what linear actuators are and how they are used.
- Explain how an electric leadscrew linear actuator works and calculate extension times.
- Understand solenoid operation and characteristics.
- Understand electric linear motor operation.
- Describe the components of a basic hydraulic system and understand how they work.
- Calculate the force generated by a hydraulic cylinder, given the system parameters.
- Describe the components of a basic pneumatic system and understand how they work.
- Calculate the force generated by a pneumatic cylinder, given the system parameters.
- Understand the basic control-valve operation.
- Understand electric linear-motor operation.

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### INTRODUCTION

The **prime mover** or **actuator** provides the source of mechanical power in a control system. It is easy to identify the prime mover in a system because it will be the source of the first physical movement. Typical prime movers are motors, hydraulic cylinders, and control valves. Technically speaking, a prime mover is a transducer as well because it converts one kind of energy into another, usually electrical energy into mechanical energy. By far the most common type of prime mover is the electric motor, which is an efficient and versatile converter of electrical to mechanical energy. However, many real-life control

applications require a slow, powerful, back-and-forth straight-line motion instead of the higher speed rotary motion produced by the electric motor. An actuator that can provide a back-and-forth straight-line motion is called a **linear actuator**.<sup>\*</sup> Many examples of linear actuators can be found in control systems, particularly in the field of robotics—for example, an object is picked up, moved in a linear fashion to another spot, and set down.

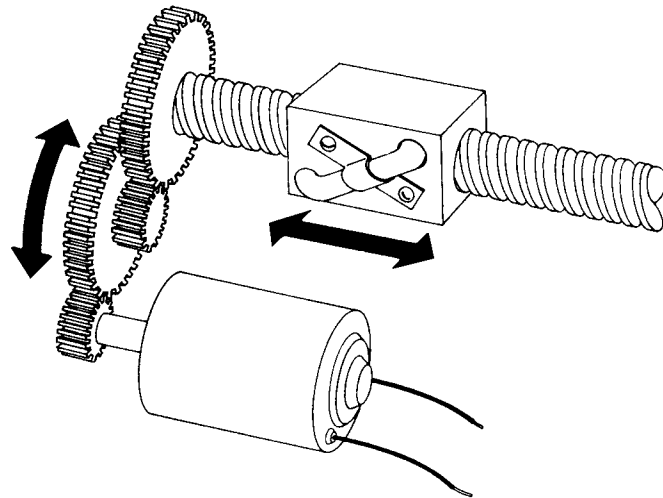
Some devices (such as the solenoid and the linear motor) can convert electrical energy directly into linear motion, but more often it is done in two stages. First, a motor is used to create rotary mechanical motion; second, the rotary motion is converted into linear motion. An example would be a motor driving a rack-and-pinion gear train (Figure 5.25), which moves the print head across the paper on your printer. Hydraulic systems are more complicated—that is, a motor drives a pump that pushes fluid into a hydraulic cylinder that causes the piston to move.

The three general types of linear actuators are electric, hydraulic, and pneumatic. Each type has its particular advantages and disadvantages, and all are in wide use. This chapter will introduce you to these systems and to flow-control valves.

## 10.1 ELECTRIC LINEAR ACTUATORS

Of the three types of linear actuators (electric, hydraulic, and pneumatic), electric actuators require the least hardware and maintenance. However, they tend to be relatively

**Figure 10.1**  
Diagram of an  
electric linear  
actuator. (Courtesy  
of Warner Electric)



<sup>\*</sup>This text uses the term linear actuator in a general sense to mean any straight-line actuator. A more particular definition for a linear actuator is “a device where there is a linear relationship between the applied control signal and the resulting displacement.”

slow-moving or to have other restrictions, which makes them suitable in only a limited number of applications. For example, electric actuators work well for lowering the landing gear in an airplane but could not be used for a spray-painting robot, which must make quick, long sweeps.

### Leadscrew Linear Actuators

A **leadscrew linear actuator** has a threaded shaft (called the *leadscrew* or *drive screw*), which is rotated by an electric motor. A “nut” on the shaft moves in a linear motion as the shaft rotates. Figure 10.1 shows a diagram of the electric linear actuator. Notice that the rotary torque of the motor is transferred through a gear train to the drive screw. The output motion results as the nut advances or retreats along the threads, depending on which direction the shaft is rotating. Note that the nut must be constrained from rotating.

To reduce the friction and increase the efficiency, many units use ball bearings to transfer the force from the drive screw to the nut. Figure 10.2 shows a cutaway diagram of this design, called a **ball bearing screw**. When the shaft rotates, the balls roll in the grooves. When the balls get to one end of the nut, they are recirculated to the other end through small pipes. Ball bearing screws practically eliminate friction when compared with the traditional nut on thread design.

**Figure 10.2**

Cutaway drawing of a ball bearing screw. (Courtesy of Warner Electric)



**Figure 10.3**

An electric linear actuator.  
(Courtesy of Warner  
Electric)



Figure 10.3 shows a cutaway diagram of a practical electric linear actuator. The unit resembles and acts like a hydraulic cylinder, the main visual difference being the attached electric motor. Many units come with a built-in electrically activated brake that prevents the shaft from rotating when the motor is not energized. This feature ensures that a heavy load cannot back drive the motor after it is turned off. Also, “over-travel” limit switches should be included in the application design, to cut off power to the motor if it attempts to drive beyond its intended travel.

Electric linear actuators come in all sizes, with rated loads from under 100 lb to over 1000 lb and strokes from 4 to 36 in. and beyond. An important parameter for electric actuators is **extension rate**, that is, how fast the unit can extend and contract. A typical extension rate is around 1 in./s, but this depends on the size of the motor and the load. However, even the larger units are no match for hydraulic cylinders (covered in Section 10.2) when it comes to speed and power. The electric actuator is characterized by slower, steady, linear movement. Figure 10.4 shows the characteristics of a 12-Vdc linear actuator. Notice that as the load increases, the extension rate decreases, and the required current increases.

### EXAMPLE 10.1

An electric linear actuator is used to lift 600 lb castings for a vertical distance of 8 in. Using the data from Figure 10.4, how long will the lifting operation take?

### SOLUTION

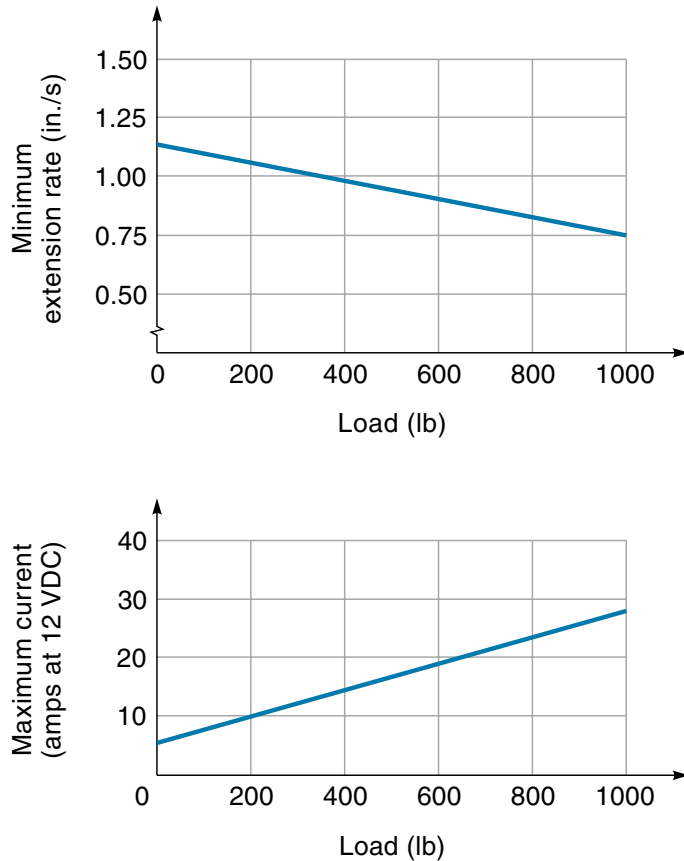
Using the top graph of Figure 10.4, we see that for a load of 600 lb the extension rate is about 0.9 in./s. The time required to lift a distance of 8 in. is calculated as follows:

$$\text{Time to lift} = 8 \text{ in.} \times \frac{\text{s}}{0.9 \text{ in.}} = 8.9 \text{ s}$$



**Figure 10.4**

Performance characteristics of a typical electric linear actuator.



For applications that can tolerate the slower extension rates, the electric linear actuator has some distinct advantages over hydraulic and pneumatic systems. Specifically, electric actuators (1) require only wires to be connected—not tubing, (2) do not require extra supporting hardware such as pumps and tanks, (3) use power only when they are actually moving, and (4) when compared with hydraulic systems, are less “messy.” On the other hand, compressed air for pneumatic systems may already be available, and for working around explosive gasses, a nonelectric actuator would be safer (no sparks!).

## Solenoids

A **solenoid** is a simple electromagnetic device that converts electrical energy directly into linear mechanical motion, but it has a very short **stroke** (length of movement), which limits its applications. The solenoid consists of a coil of wire with an iron plunger that

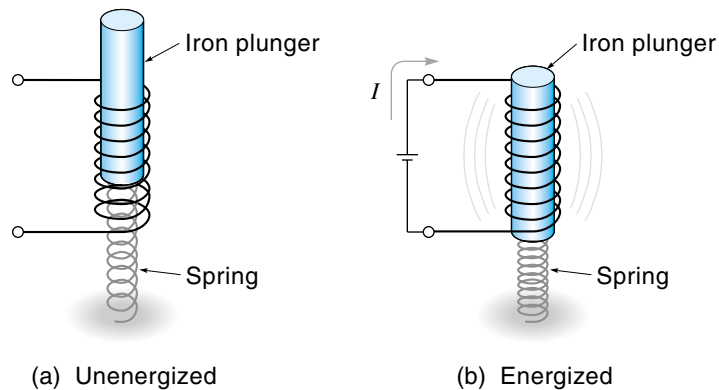
is allowed to move through the center of the coil. Figure 10.5(a) shows the solenoid in the unenergized state. Notice that the plunger is being held about halfway out of the coil by a spring. When the coil is energized [Figure 10.5(b)], the resulting magnetic field pulls the plunger to the middle of the coil. The magnetic force is unidirectional—a spring is required to return the plunger to its unenergized position. Figure 10.6 shows two practical solenoids.

The main limitation of the solenoid is its short stroke, which is usually under an inch. Still, there are many applications for short-stroke linear motion; examples are activating electric car-door locks, opening and closing valves, and triggering mechanical latches. Most applications use the solenoid as a on or off device—that is, the coil is either completely energized or switched off. However, variable-position control is possible by varying the input voltage.

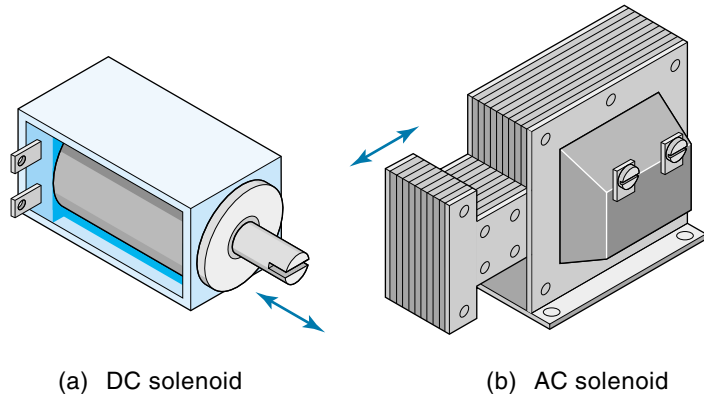
Both AC and DC solenoids are used, the major difference being that AC solenoids use a plunger and frame made from laminations instead of solid iron. *Laminations* are thin sheets of lacquered iron that are riveted together to form the frame and plunger. Laminations prevent power-consuming *eddy currents* (induced by the AC) from circulating in the metal parts of the solenoid. The laminations can easily be seen in Figure 10.6(b).

Table 10.1 lists various models of solenoids. Note the various coil voltages and whether the unit can be operated intermittently or continuously. An *intermittent-duty solenoid* is designed to operate for a short time and then take time to cool. The problem is that, like relays, it takes more voltage to pull in the solenoid than it takes to hold it in the retracted position. Thus, if a solenoid is left on, it is drawing more current than it really needs, and it tends to heat up. A *continuous-duty solenoid* is designed to deal with the heat, so it can operate all the time. The “Lifts and Strokes” data in Table 10.1 tell us the maximum pull force (which occurs when the plunger is retracted). The minimum pull force occurs when the plunger is extended. For example, the unit on the first line of Table 10.1 exerts (at least) 8 oz of force for the first  $\frac{1}{8}$  inch of stroke but falls to only 2 oz of force when the solenoid is extended to its maximum of  $\frac{1}{2}$  inch.

**Figure 10.5**  
A solenoid.



**Figure 10.6**  
Solenoids.



**TABLE 10.1**

**A Selection of Solenoids**

Type	Volts	Duty	Ohms	Lifts and Strokes (ounces at inches)		Body
				(minimum)	(maximum)	
870	120 AC	Inter.	300	8 @ 1/8	2 @ 1/2	1
871	120 AC	Cont.	675	3 @ 1/8	0.5 @ 7/8	1
078	120 AC	Inter.	60	33 @ 1/8	9 @ 7/8	1
079	120 AC	Cont.	166	10.5 @ 1/8	3 @ 7/8	1
1897	120 AC	Inter.	36	83 @ 1/8	16 @ 3/4	1
1898	120 AC	Cont.	113	17 @ 1/8	4 @ 3/4	1
1899	24 DC	Inter.	20	100 @ 1/8	5 @ 3/4	1
1900	24 DC	Cont.	64	50 @ 1/8	1 @ 5/8	1
872	120 AC	Inter.	37	24 @ 1/8	22 @ 1	2
873	120 AC	Cont.	133	8 @ 1/8	6 @ 1	2
885	24 DC	Inter.	15	110 @ 1/8	4 @ 1	2
886	24 DC	Cont.	54	55 @ 1/8	2.5 @ 7/8	2
887	115 DC	Inter.	346	90 @ 1/8	4 @ 1	2
888	115 DC	Cont.	1300	55 @ 1/8	2.5 @ 7/8	2
1901	24 DC	Inter.	15	156 @ 1/8	21 @ 3/4	2
1902	24 DC	Cont.	50	70 @ 1/8	4 @ 3/4	3
1903	115 DC	Inter.	350	156 @ 1/8	21 @ 3/4	3
1904	115 DC	Cont.	1200	70 @ 1/8	4 @ 3/4	3
5600	120 AC	Inter.	220	17 @ 1/8	8 @ 1/2	4
5601	120 AC	Cont.	350	11 @ 1/8	4 @ 1/2	4
5602	24 DC	Inter.	52	16 @ 1/8	1 @ 1/2	4

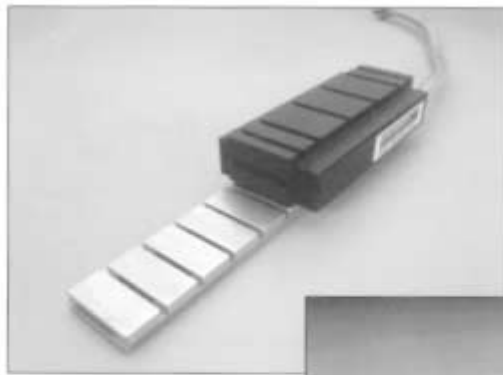
## Electric Linear Motors

One of the newer types of linear actuators is the electric linear motor. Basically, a **linear motor** is a rotary brushless DC motor (BLDC) that has been rolled out flat. Linear motors convert electric power directly into linear motion. They are capable of high speed (up to 10 m/s), high force (over 1000 lb) and long travel (several meters), and, like a BLDC, they have no sliding contacts but do require an electronic drive unit.

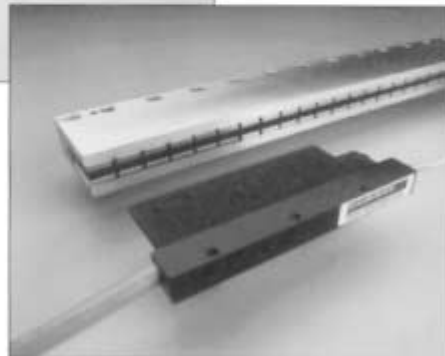
A linear motor consists of two parts: the base unit known as the *magnet way*, which consists of an iron plate with a row of alternate-pole permanent magnets, and the moving part known as the *slider*, which contains the coils. These parts are shown in Figure 10.7. The slider is supported on linear ball bearings so that it can slide along over the magnet way (there is a small air gap between them). A flexible cable connects the slider to the drive unit.

Figure 10.8 illustrates the operation of a simplified three-phase linear motor. Note that there are three coils spaced over four magnets. In Figure 10.8(a), coil A is switched on, causing it to become a north pole (N). The attractive force between this N pole and the nearest S pole on the magnet way causes the slider to move to the right. When the slider has moved to the position shown in (b), coil A is switched off and coil B is switched on, causing renewed attractive force to the right, and the slider moves further (to the right). When the slider gets to position (c), coil B is switched off and coil C is switched on, causing the slider to move to position (d). The coils have now advanced

**Figure 10.7**  
Electric linear  
motors. (Courtesy  
of Kollmorgen)



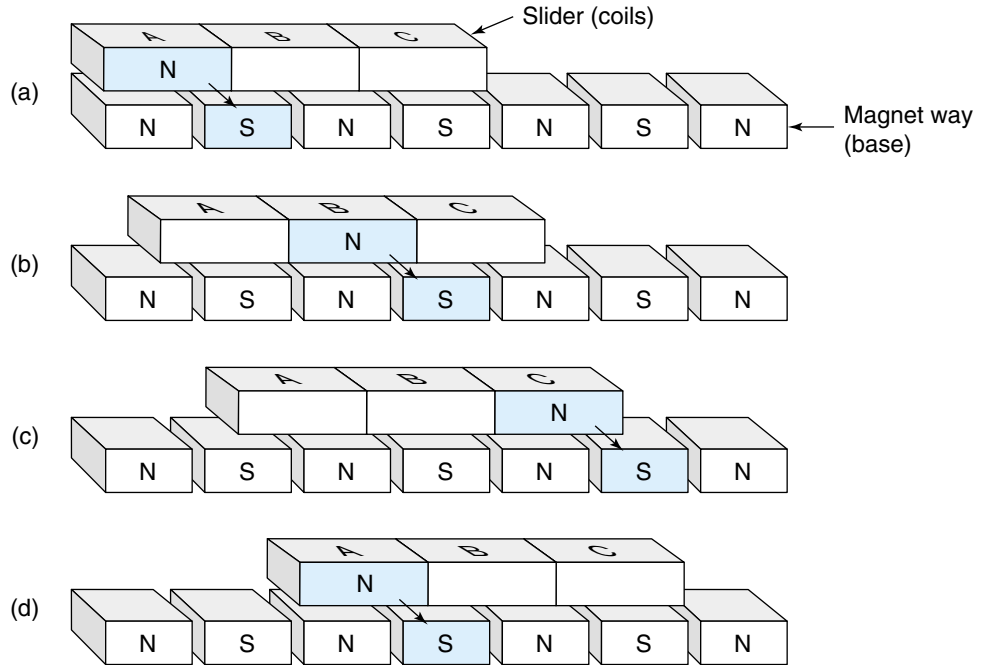
(a) Iron-core motor



(b) Ironless motor

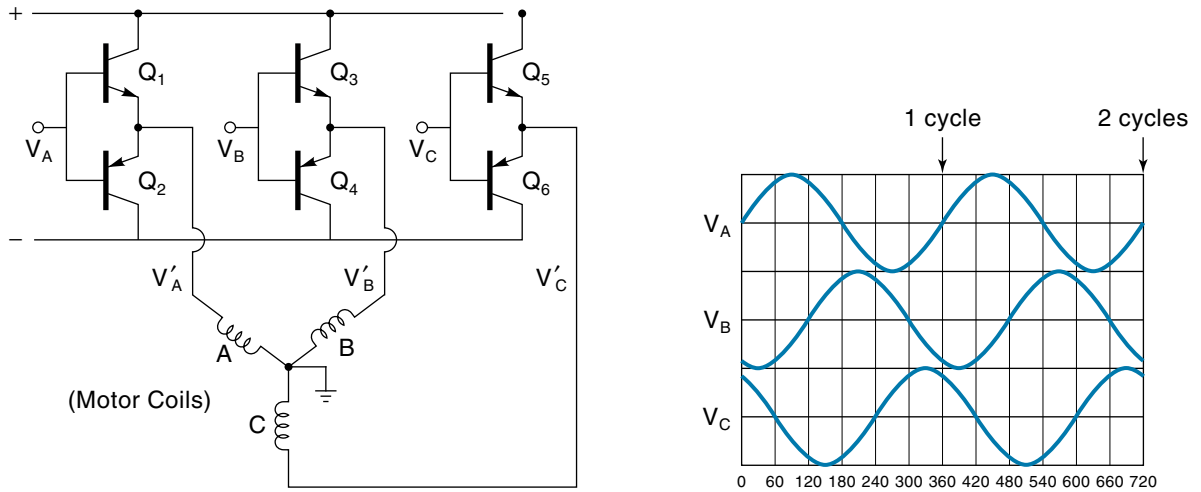
**Figure 10.8**

Operation of a simplified three-phase linear motor.



one full cycle. They are now back to the same orientation as in (a), and the whole sequence starts over. As you might imagine, moving the slider in steps tends to cause a jerky motion known as **cogging**. Cogging is more noticeable at lower speeds and can be greatly reduced by proper shaping of the coil drive currents (see the next paragraph). Switching the coils on and off at the right time in the right sequence without brushes is called *electronic commutation*. For this example, electronic commutation could consist of three Hall-effect sensors mounted along the slider, each one controlling power to a coil. The Hall-effect device detects when the slider has physically moved to the next position and signals the drive unit to energize the next coil.

An actual linear motor would probably be operated in a bipolar mode, wherein each coil becomes alternately north and south, and two of the three coils are on at the same time. This system produces smoother operation with more torque but requires a more complicated drive unit because current reversals are necessary. Figure 10.9(a) shows one schematic for a three-phase bipolar linear motor that permits the current in each coil to go in both directions (allowing it to be either a north or a south pole). The three push-pull amplifiers ( $Q_1$ – $Q_2$ ), ( $Q_3$ – $Q_4$ ), ( $Q_5$ – $Q_6$ ) could be driven by the three waveforms shown in Figure 10.9(b). These waveforms would be created from the Hall-effect sensor outputs. Notice in Figure 10.9(b) that the voltage peaks in the sequence A, B, C; this would correspond to energizing the coils as a “north” in the sequence A, B, C. Using sine waves to turn on the coils causes the force to be passed gently from one coil to the next and practically eliminates cogging.



(a) Schematic for three-phase operation

(b) Phase voltage sequence

**Figure 10.9**

Driver circuit for a three-phase linear motor.

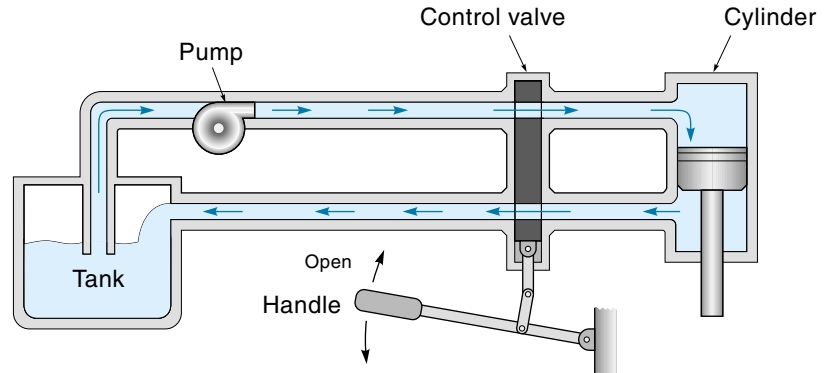
There are two basic designs for linear motors, the *iron-core linear motor* and the *ironless linear motor*; as shown in Figure 10.7. Iron-core motors (the subject of the preceding paragraphs) are more powerful because they have coils wound on steel laminations to maximize their magnetic force. However, in addition to the linear force, they also generate a strong attractive force between the slider and the magnet way. This attractive force is typically four times the linear force, so a motor pulling with 100 lb is pushing down on the magnet way with 400 lbs. The ironless linear motor has no iron at all in the slider, just copper wires embedded in an epoxy strip. These motors are inherently weaker, even when two sets of magnets are used in the magnet way, but they have small mass (for quicker acceleration), no cogging, and no attractive forces.

Linear motors are almost always used in a closed-loop system, which means they need some kind of linear position sensor. A common and accurate type of linear position sensor is an optical encoder; it is similar to an optical shaft encoder rolled out flat (optical shaft encoders are discussed in Chapter 6). The components of this sensor are a yardstick-like strip (marked with fine vertical lines or slots) fastened on the slider and a photo sensor mounted to the base. In some cases, the linear motor uses the position sensor data for electronic commutation, eliminating the need for Hall-effect sensors.

## 10.2 HYDRAULIC SYSTEMS

**Hydraulic systems** are ideally suited to provide a strong, fast, or slow linear motion. The system uses a fluid (light-grade oil) to transfer energy from a pump to an actuator.

**Figure 10.10**  
A simplified  
hydraulic system.



A simplified system, illustrated in Figure 10.10, consists of the following components: a tank of hydraulic fluid, a pump, a control valve, and a cylinder. The pump pushes the fluid through a tube to the control valve. The control valve directs the fluid to the cylinder, causing the piston to move down in response to the fluid pressure. The pump is the actual source of mechanical power, and it is physically separate from the cylinder actuator. The practical significance of this fact is that *only the cylinder needs to be mounted at the place where the motion is needed*; the pump can be elsewhere. This allows a relatively small component such as a hydraulic cylinder to provide far more power than a similarly sized electric actuator, which must have the motor attached.

### Basic Principles of Hydraulics

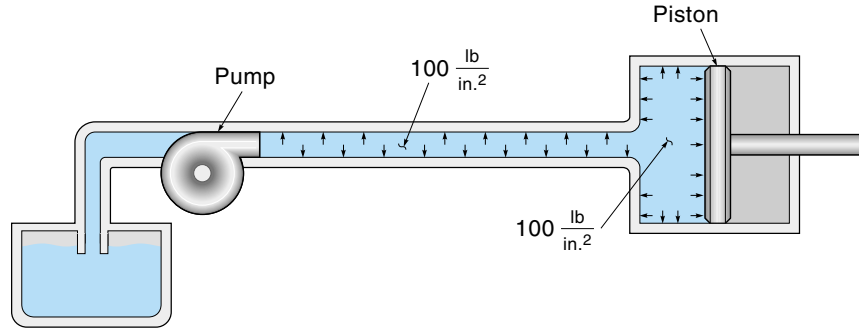
One major difference between a liquid and a gas is that a gas will squeeze down into a smaller volume when pressurized, whereas a liquid will not. We say that liquids (including hydraulic fluid) are incompressible. This is an important concept to remember when studying hydraulic systems. It means that each cubic inch of fluid pumped into the system must appear someplace else downstream as  $1 \text{ in}^3$  of fluid volume.

Another important concept is that of **hydrostatic pressure**, which results when a fluid is under pressure and it is not moving (hence, static). Pascal's principle states that *a fluid under hydrostatic pressure will exert the same pressure uniformly on the inside walls of all the interconnected components*. Recall that pressure is defined as a "force per unit area." For example, a pressure\* of  $100 \text{ lb/in}^2$  (psi) means that the container must withstand a force of 100 lb for every square inch of surface area. The unit of pressure for SI units is the Pascal (Pa), where  $1 \text{ Pa} = 1 \text{ Newton/meter}^2$  ( $\text{N/m}^2$ ). This pressure is so small that a more common unit is the kiloPascal ( $1000 \text{ Pa} = 1 \text{ kPa}$ ). The relationship between psi and kPa is  $1 \text{ psi} = 6.89 \text{ kPa}$ .

\*In this chapter, the pressure values referred to are technically called *gauge pressure*, meaning "that pressure above the ambient atmospheric pressure of 14.7 psi or  $10^5 \text{ Pa}$ ."

**Figure 10.11**

Hydrostatic pressure pushes out in all directions.



The principle of hydrostatic pressure is illustrated in Figure 10.11, which shows a tank, pump, piston, and hydraulic cylinder. The fluid on the high-pressure side of the pump is pressurized to 100 psi. This means that the tubing, cylinder walls, and piston face are all receiving 100 lb of force for each square inch of area, as indicated by the arrows in Figure 10.11. All internal surfaces are under pressure, but only the piston is movable, so it moves to the right. The force on the piston from the hydrostatic pressure is the product of the pressure times the area of the piston face, as given in Equation 10.1:

$$F = PA \quad (10.1)$$

where

$F$  = total force exerted on the piston

$P$  = hydrostatic pressure

$A$  = area of the piston face

**EXAMPLE 10.2**

The hydraulic cylinder in Figure 10.11 is 3 in. in diameter. Find the force exerted on the piston if the pressure is 100 psi.

**SOLUTION**

We can use Equation 10.1 to find the force, but first we need to find the area of the piston face:

$$\text{Area} = \pi r^2 = 3.14 \times (1.5 \text{ in.})^2 = 7.07 \text{ in}^2$$

Now we can calculate the piston force, using Equation 10.1, knowing that the pressure is 100 psi:

$$\text{Force} = PA = \frac{100 \text{ lb}}{\text{in}^2} \times 7.07 \text{ in}^2 = 707 \text{ lb}$$



**10.2 (Repeated with SI Units)**

The hydraulic cylinder in Figure 10.11 is 8 cm in diameter. Find the force exerted on the piston if the pressure is 700 kPa.

**SOLUTION**

We can use Equation 10.1 to find the force, but first we need to find the area of the piston face:

$$\text{Area} = \pi r^2 = 3.14 \times (4 \text{ cm})^2 = 50.24 \text{ cm}^2$$

Now we can calculate the piston force, using Equation 10.1, knowing that the pressure is 700 kPa:

$$700 \text{ kPa} = 700,000 \text{ Pa} = 700,000 \text{ N/m}^2$$

$$\text{Force} = PA = \frac{700,000 \text{ N}}{\text{m}^2} \times \frac{1 \text{ m}}{100 \text{ cm}} \times \frac{1 \text{ m}}{100 \text{ cm}} \times 50.24 \text{ cm}^2 = 3517 \text{ N}$$

**EXAMPLE 10.3 (Mixed English and SI Units)**

A 2-in. diameter hydraulic cylinder is being supplied with a pressure of 1000 kPa. Find the force exerted on the piston in pounds.

**SOLUTION**

We can use Equation 10.1 to find the force, but first we need to find the area of the piston face:

$$\text{Area} = \pi r^2 = 3.14 \times (1 \text{ in.})^2 = 3.14 \text{ in}^2$$

Now we can calculate the piston force, using Equation 10.1, but first we must convert the pressure to psi:

$$1000 \text{ kPa} \times \frac{1 \text{ psi}}{6.89 \text{ kPa}} = 145.1 \text{ psi}$$

$$\text{Force} = PA = \frac{145.1 \text{ lb}}{\text{in}^2} \times 3.14 \text{ in}^2 = 455.6 \text{ lb}$$

Static hydraulic systems can be used to “amplify” a force. This is the principle used in hydraulic jacks or hydraulic brake systems and is illustrated in Figure 10.12. The system contains two cylinders of different diameters that are connected by a tube. When a force is applied to the small piston, a pressure develops. This same pressure is transmitted through the tube and is applied to the large piston. Because the large piston has more surface area, it receives a larger net force. Of course, it is also true that the large piston will move a shorter distance than the small piston because the work done by the two pistons must be the same.

#### EXAMPLE 10.4

For the system shown in Figure 10.12, the small piston has a face area of  $2 \text{ in}^2$  and receives an external force of  $10 \text{ lb}$ . The large piston has a face area of  $20 \text{ in}^2$ . Calculate the force exerted by the large piston.

#### SOLUTION

The small piston has a surface area of  $2 \text{ in}^2$  and is being pushed with a force of  $10 \text{ lb}$ . Rearranging Equation 10.1 ( $F = PA$ ) and solving for  $P$ , we can calculate the fluid pressure this will create:

$$P = \frac{F}{A} = \frac{10 \text{ lb}}{2 \text{ in}^2} = \frac{5 \text{ lb}}{\text{in}^2}$$

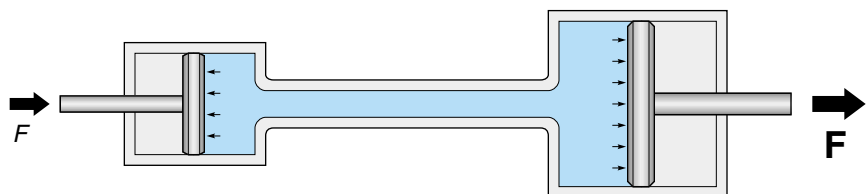
This pressure of  $5 \text{ psi}$  is conveyed to the large cylinder where it pushes on the  $20\text{-in}^2$  piston. Rearranging Equation 10.1 again, we can calculate the force on the large piston:

$$F = PA = \frac{5 \text{ lb}}{\text{in}^2} \times 20 \text{ in}^2 = 100 \text{ lb}$$

Thus, the large piston will exert a force ten times larger but will move only one-tenth the distance of the smaller piston because the *volume change must be the same for both sides*.

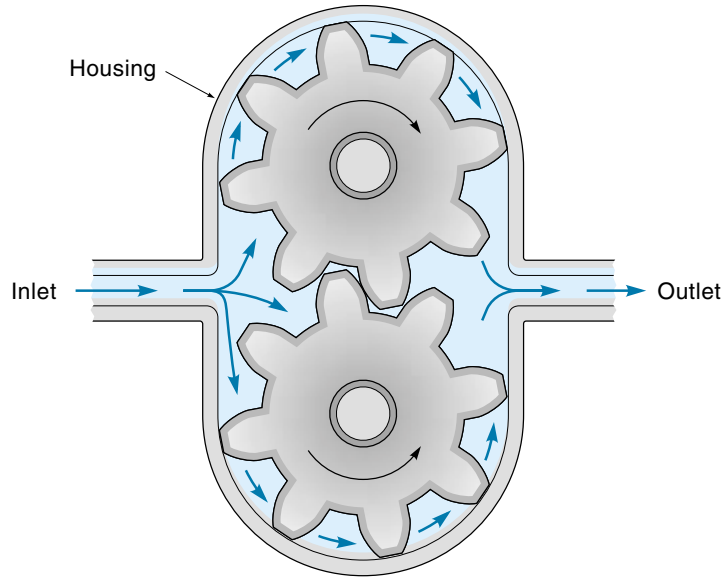
**Figure 10.12**

A hydraulic system that “amplifies” force.



**Figure 10.13**

A hydraulic gear pump.



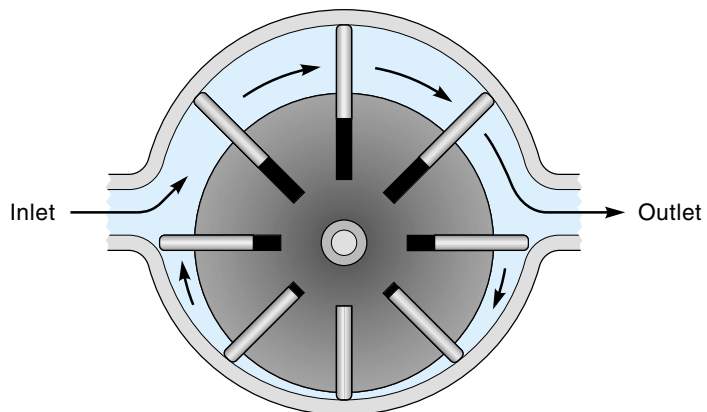
### Hydraulic Pumps

In an active hydraulic system, a pump is used to create the hydrostatic pressure. Figure 10.13 shows a common pump design called a **gear pump**, which consists of two meshed gears in a housing. As the gears rotate, fluid is trapped in the little spaces between the teeth and the housing (both top and bottom) and is conveyed from the inlet to the outlet. The mesh between the gears in the center is tight enough so that no fluid moves through either way at that point. This type of pump is also known as a **positive-displacement pump** because a constant volume of fluid is pumped for every revolution of the gears.

The **vane pump** is another type of hydraulic pump. Shown in Figure 10.14, it consists of an offset rotor in a housing with retractable vanes. The spring-loaded vanes push

**Figure 10.14**

A hydraulic vane pump.



out and seal against the housing wall. Because there is more fluid between the vanes in the top half of the housing than in the bottom, there is a net transfer of fluid from the inlet to the outlet. In some designs, the position of the rotor axis is adjustable. The more offset the rotor axis, the more fluid is pumped. Such a pump is called a **variable-displacement pump**.

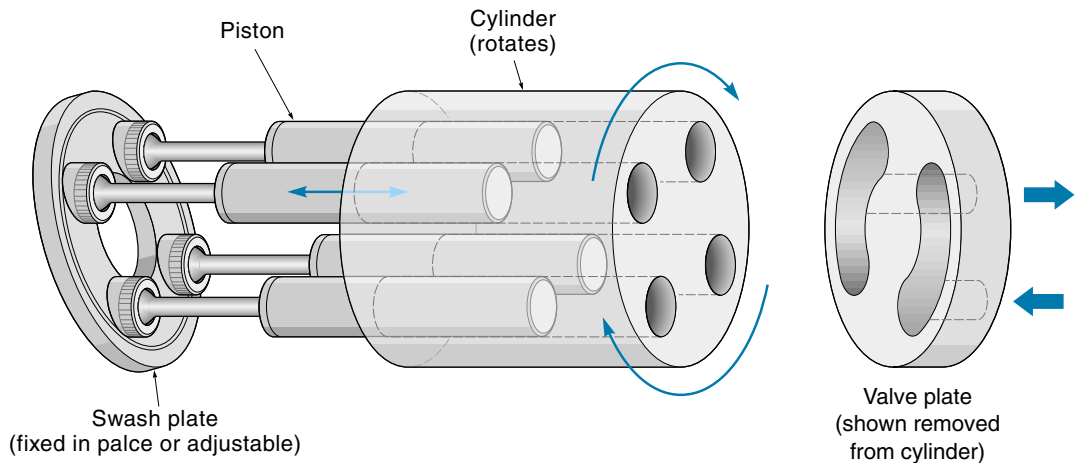
A third kind of pump, an **axial piston pump**, uses small pistons reciprocating back-and-forth to pump the fluid. As shown in Figure 10.15, the pump consists of a rotating cylinder and a metal ring called a **swash plate** (which does not rotate). The cylinder contains a number of small pistons that do the actual pumping. One end of each piston rides against the swash plate. Because the swash plate is at an angle to the cylinder, each piston is forced to move in-and-out with each rotation of the cylinder. By changing the angle of the swash plate, the quantity of fluid pumped per revolution is changed.

### Hydraulic Actuators

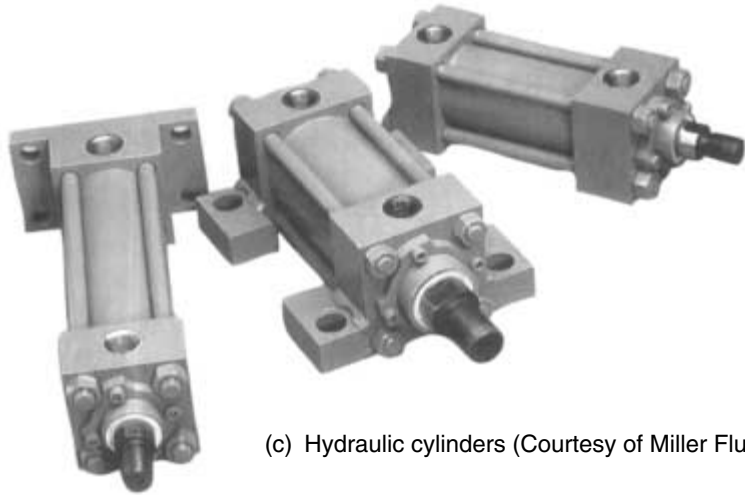
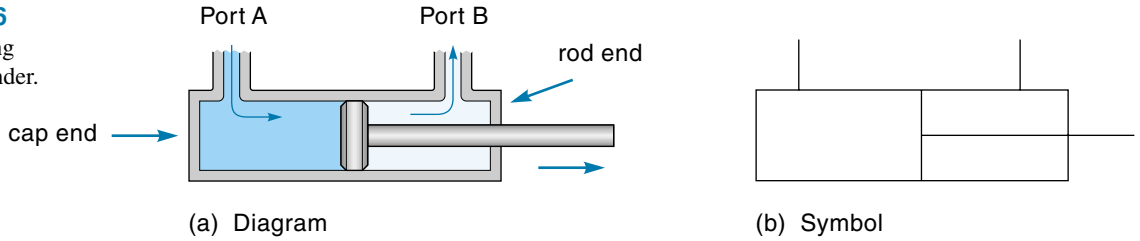
The most common type of hydraulic actuator is the **hydraulic cylinder**. A typical cylinder is shown in Figure 10.16(a) and is known as a **double-acting cylinder** because it can provide force in either direction. It consists of a piston and a cylinder body. The piston has a rod that extends out one end of the cylinder. Fluid can enter and leave the cylinder on either side of the piston through ports. Under normal operating conditions, both ends of the cylinder are filled with fluid. If additional fluid enters port A, the piston will move toward the right, but the fluid must be able to escape through port B. A selection of hydraulic cylinders is shown in Figure 10.16(c).

Some hydraulic actuators can create rotary motion and are very similar to the pump designs discussed previously. For example, the **gear motor** rotary actuator shown in Figure 10.17 is almost identical to the gear pump. For the motor, fluid is pumped in the left side of the case, putting that area under pressure (indicated by  $p$ ). Within the pressurized area, all surfaces receive a force, but only those three surfaces indicated with arrows will affect rotation (the other surfaces are balanced out or unmoveable). The

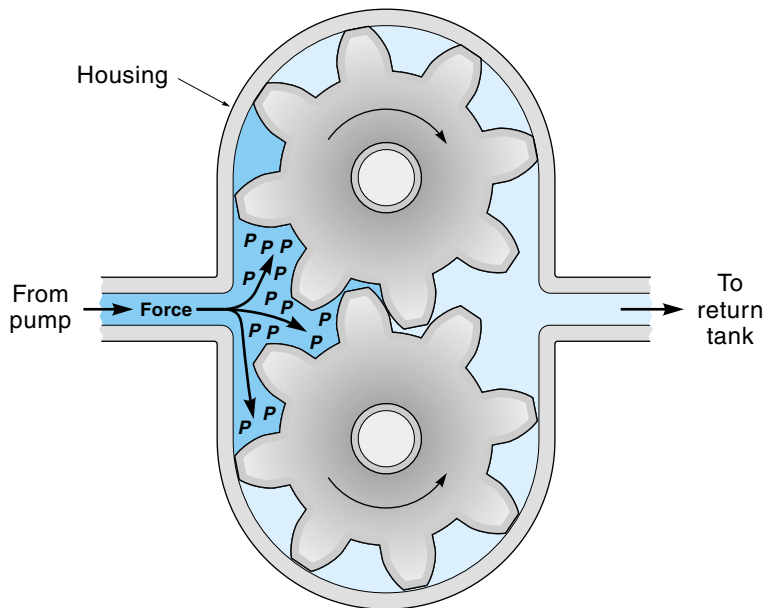
**Figure 10.15**  
An axial piston pump.



**Figure 10.16**  
A double-acting hydraulic cylinder.

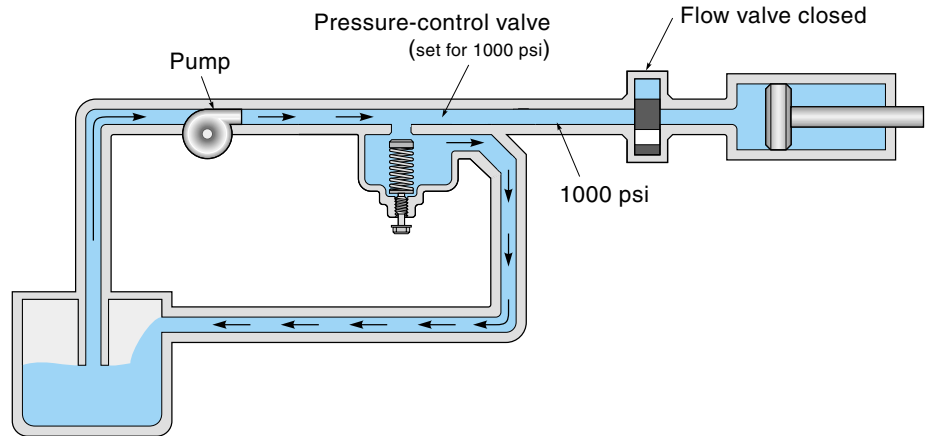


**Figure 10.17**  
A hydraulic gear motor.



**Figure 10.18**

A relief valve circulates surplus fluid to the tank.



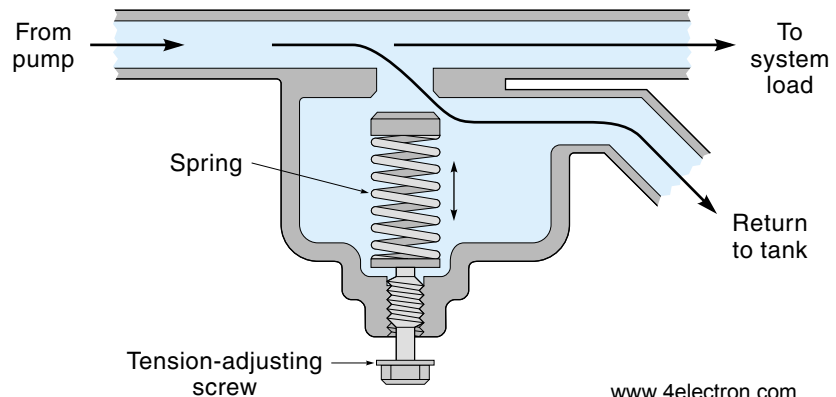
pressure on the teeth next to the case (top and bottom arrows) will cause the gears to rotate as shown. The pressure on the meshing teeth in the center would cause the gears to turn in the opposite direction, but this torque is overpowered because two teeth (top and bottom) are pushing the other way.

### Pressure-Control Valves

A **pressure-control valve** is a spring-loaded valve that is capable of maintaining a constant pressure in a system, regardless of the flow rate. This is important because most pumps (such as the gear pump) are constant-displacement types—a constant volume of fluid is pumped for each revolution of the pump shaft. If the pump were connected directly to the cylinder, it would have to start and stop each time the piston moved to a new position. Such on-off cycling reduces the lifetime of machinery and is therefore undesirable. When a pressure-control valve is put into the system (as shown in Figure 10.18), the pump can remain on the whole time—when the fluid pressure exceeds the

**Figure 10.19**

A pressure-control valve (partially open position).



preset limit, the valve opens, and the surplus fluid is simply returned to the tank. In other words, when the piston is *not* moving, the fluid is simply circulating from the tank, through the pump, through the valve, and back to the tank. If the pressure-control valve opens at 1000 psi, then the pressure in the lines will never get much above 1000 psi.

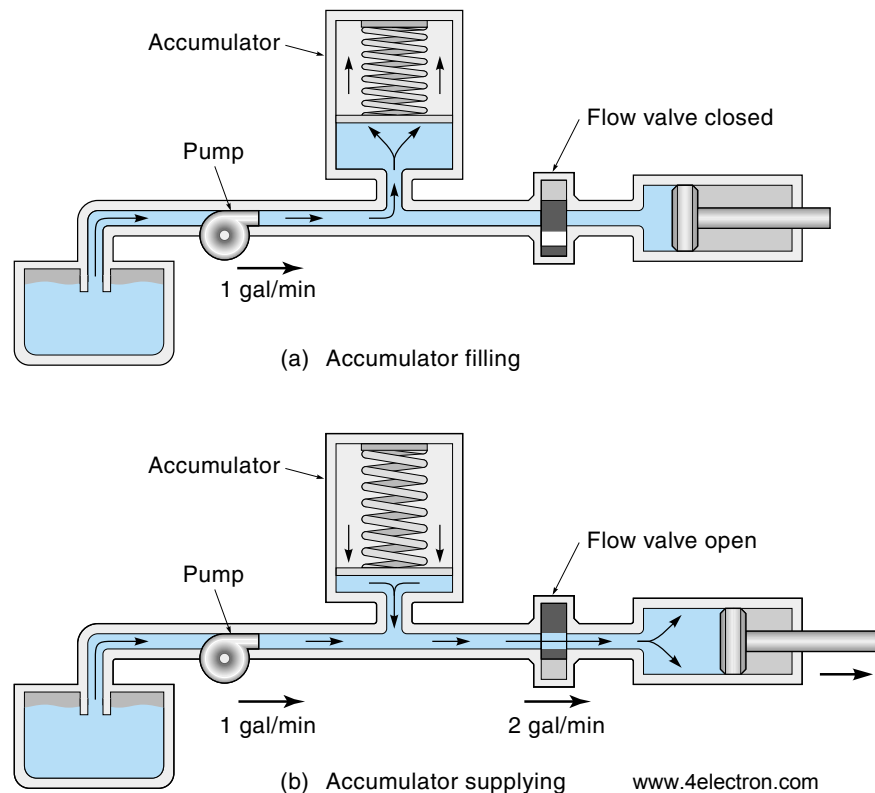
Figure 10.19 shows the inside of a simple pressure-control valve. Notice that the passage from the pump to the load is always open. If the hydraulic pressure in the main passage overcomes the spring pressure, the valve opens, and fluid can escape out a separate port that is connected back to the tank. The pressure at which the valve opens can be set by adjusting the tension on the spring.

## Accumulators

An **accumulator**, which is connected into the system, is a special kind of spring-loaded storage tank for hydraulic fluid (Figure 10.20). The accumulator serves two functions. First, it acts as a low-pass filter to remove pressure pulsations from the pump; second, it stores extra fluid for those high-demand times when the actuator requires fluid at a faster rate than the pump can supply. This latter situation is illustrated in Figure 10.20, which shows a pump (capacity: 1 gal/min) supplying fluid to a cylinder. Notice that the accumulator is located between the pump and the cylinder. In Figure 10.20(a) the valve to the

**Figure 10.20**

An accumulator in a system. (For simplicity, pressure regulator and return line are not shown.)



cylinder is closed; consequently, all fluid from the pump goes into the accumulator, compressing the accumulator spring. In Figure 10.20(b), the valve opens enough to allow fluid to the cylinder at a rate of 2 gal/min. Because the pump can supply only 1 gal/min, the other 1 gal/min comes from the stored fluid in the accumulator, which is pushed out by the spring. Naturally, this concept will work only if the cylinder's demand for fluid is intermittent, but this is often the case. In fact, this is one of the advantages of a hydraulic system; it can store energy within the system, to be used during periods of peak demand. For example, consider the case where a robot performs a 10s lifting operation (which requires 1 hp) once each minute. Because the average load is only  $\frac{1}{6}$  hp [ $1 \text{ hp} \times 10 \text{ s}/60 \text{ s}$ ], a hydraulic system could get by with a  $\frac{1}{6}$ -hp pump. For 50 s out of each minute, the pump would be "charging up" the accumulator; for 10 s, both the pump and the accumulator would be supplying the cylinder. Now suppose, instead of the hydraulic system, the robot uses an electric linear actuator. Because the electric motor cannot store up energy, it would have to be the full 1 hp, even though it is only on one-sixth of the time.

The three types of accumulators are: spring-loaded (already mentioned), weighted, and gas-pressurized. The weighted type uses a weight to provide the pressure on the fluid. This has the advantage of providing a constant pressure but the disadvantage of needing to be mounted in an upright and relatively stable position. The gas-pressurized type uses a gas (dry nitrogen) under pressure to provide the push on the fluid. This is the most popular type because the pressure it exerts is easily adjustable and it can be mounted in any position.

## Directional Control Valves

The flow-control valve is used to regulate the flow rate of hydraulic fluid. For this discussion, the most important type of flow valve is the **directional control valve**, which regulates the movement of the piston in the cylinder. Figure 10.21 shows a diagram of this type of valve. A sophisticated device, it must be capable of admitting pressurized fluid to either end of the cylinder while providing a return path for fluid being squeezed out of the other end of the cylinder.

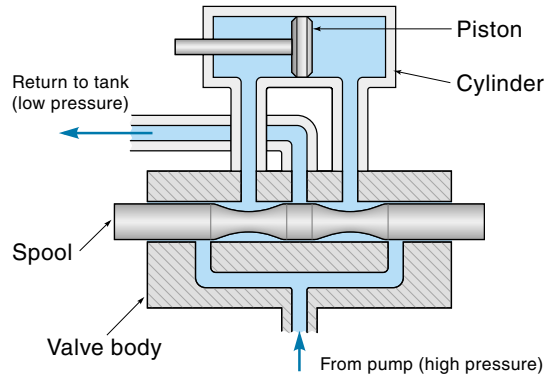
The control valve consists of two major parts: the valve body and the spool. Referring to Figure 10.21, you can see that the valve body has four fluid-connection ports: (1) high-pressure fluid supply (from the pump), (2) low-pressure return to the tank, and (3 and 4) connections to both ends of the cylinder. The **spool** is a solid-metal machined shaft that can slide back-and-forth in the valve body. The spool in Figure 10.21 has two deep grooves that allow fluid to pass from one port in the valve body to another. The position of the spool within the valve body determines how fast, and in what direction, the piston will move. The spool is moved by a hand lever, an electric actuator such as a solenoid, or a small hydraulic or pneumatic actuator.

The following is a description of the operation of the spool-type valve of Figure 10.21. (To really understand this clever design, take the time as you read to trace the flow of fluid in each diagram.) Figure 10.21(a) shows the valve with the spool centered. In this position, the fluid from the pump is completely blocked by the spool.

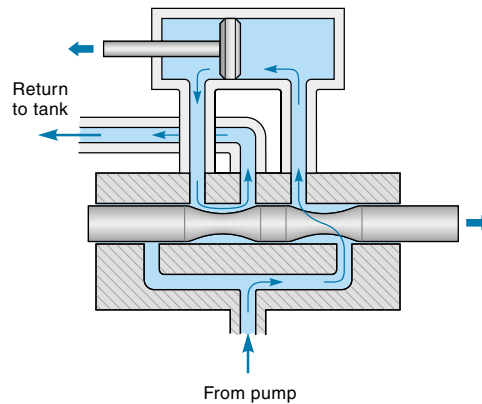


**Figure 10.21**

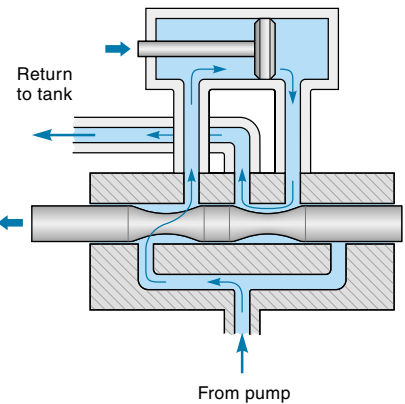
A directional control valve for a double-acting cylinder.



(a) Spool centered



(b) Spool to right



(c) Spool to left

The lines connected to the cylinder are also blocked; with the cylinder fluid thus trapped, the piston is locked in place. This explains why a backhoe hydraulic shovel will remain in the upright position even with a heavy load and the power off. (However, don't trust your safety to this feature—hoses can leak!)

Figure 10.21(b) shows the same valve with the spool moved to the right. Notice that the right-hand groove is allowing the fluid to pass from the pump to the right end of the cylinder, causing the piston to move to the left. The left spool groove has moved into position to allow an escape route for the fluid being pushed out of the left end of the cylinder. This low-pressure fluid is allowed to return to the tank.

Figure 10.21(c) shows the same valve with the spool moved to the left of center. Now the high-pressure fluid from the pump is allowed to pass, via the left groove, to the left end of the cylinder (causing the piston to move to the right). The return fluid from the cylinder passes through the right groove and then back to the tank.

**Figure 10.22**  
A basic hydraulic system (standard symbols).

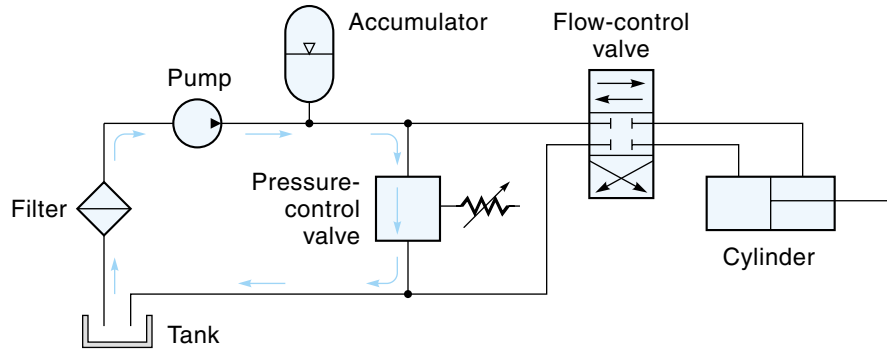


Figure 10.22 shows a diagram (using standard symbols) of a complete simple hydraulic system, which includes the tank, filter, pump, accumulator, pressure-control valve, directional control valve, and cylinder. The constant-displacement pump would be running all the time the system is on. During those times when the cylinder is not moving, the fluid from the pump, after filling the accumulator, returns to the tank through the pressure-control valve. Of special importance is the filter that removes small contaminants, which can get into the fluid. These contaminants can cause abrasive wear on the system components and reduce their lifetime considerably.

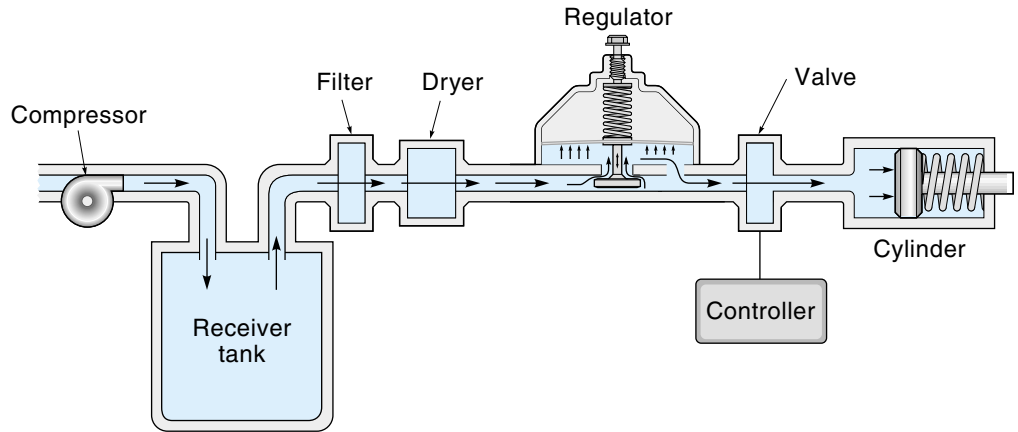
## 10.3 PNEUMATIC SYSTEMS

**Pneumatic systems** use air pressure to create mechanical motion. As shown in Figure 10.23, the basic system includes an intake filter that traps dirt before it enters the system, an air compressor that provides a source of compressed air, a dryer that removes the moisture in the air, a pressure tank that is a reservoir of compressed air, a pressure regulator that maintains air pressure, a valve that controls the air flow, and a pneumatic cylinder that creates the mechanical motion.

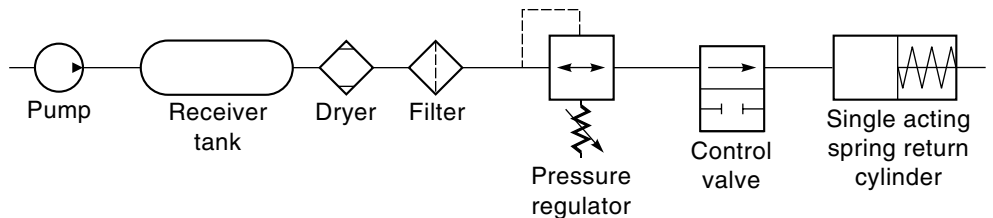
Pneumatic systems are very similar to hydraulic systems, but there are several important differences. The major functional difference is that air is compressible, whereas hydraulic fluid is not. In fact, air is so compressible that the air in the average car tire would occupy three times its volume at atmospheric pressure. To explore this concept further, consider a bicycle pump being used to inflate a tire. When you push down on the plunger, the air in the pump is compressed until the air pressure in the pump is greater than the pressure in the tire, at which point some portion of the air flows out of the pump (through a one-way check valve) into the tire. If you were to push the plunger down only halfway and let go, the compressed air in the pump cylinder would expand and push the plunger part way back out. If for some strange reason you were pumping an incompressible fluid (such as water) into your tire, you would notice a *big*

**Figure 10.23**

A basic pneumatic system (simplified).



(a) Diagram



(b) Symbolic diagram

difference. Pushing the plunger halfway down would not compress the water volume, so exactly half of the water in the cylinder would be expelled into the tire. And if you were to let go of the plunger in midstroke, the plunger would not rebound because water under pressure does not expand. The point of this analogy is that a pneumatic system tends to be springy, whereas a hydraulic system is rock solid. That is, the position of a hydraulic cylinder is strictly a function of the volume of fluid in the cylinder, whereas the position of a pneumatic cylinder is a function of the air pressure, which in turn depends on the resistance of the mechanical load, and the temperature. For example, if a hydraulic actuator is moved to position and stopped, the trapped fluid on each side of the piston will lock it in place; if a pneumatic actuator is moved into position and stopped, the piston *can* be displaced by pushing or pulling on it with enough force—the trapped air in the cylinder simply compresses somewhat. For this reason, pneumatic systems are not usually used when accurate feedback positioning is required; instead, positioning is achieved by having the piston travel back-and-forth between two mechanical stops, and enough air pressure is provided to guarantee that the piston can move any expected load.

Another difference between hydraulic and pneumatic systems is that pneumatic systems do not have to return the low-pressure air to the compressor: They simply

exhaust it to the atmosphere—clearly an advantage. Another advantage is that pneumatic systems are not “messy.” Hydraulic systems can and do leak; pneumatic systems also leak, but there are no spills to clean up. Also, pneumatic systems tend to be smaller and less expensive. All these qualities make them very attractive as a source of mechanical motion in cleaner, high-tech industrial environments.

A control system that uses pneumatic actuators would include a controller along with the pneumatic components shown in Figure 10.23. The controller specifies when the individual valves in the system are to be turned on and off. Usually, this controller is a digital electronic device such as a microcomputer or a programmable logic controller (PLC). However, some controllers are strictly pneumatic devices and use no electronics at all. Pneumatic controllers make use of pneumatic logic devices—actual AND, OR, and NOT gates that operate only on air pressure. The output signals from these controllers are, of course, pneumatic and are conveyed to the control valves through small tubes. Although pneumatic controllers are still used extensively in some applications (such as simple sequencers or around flammable gases), in general, industry is backing away from this concept.

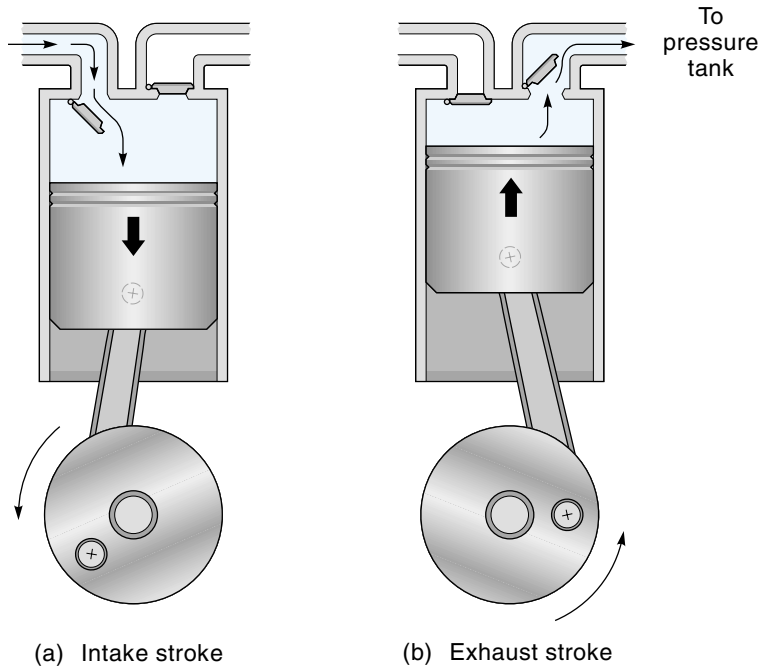
## Compressors, Dryers, and Tanks

The **air compressor** is a machine that pumps air from the atmosphere into a tank. There are a number of types of compressors, but one of the most common is the **reciprocating piston compressor** shown in Figure 10.24. The compressor crankshaft is driven by an external power source, typically an electric motor. As the crankshaft rotates, the piston is forced up-and-down in the cylinder. The air is drawn into the cylinder through a valve during the intake stroke [Figure 10.24(a)]; then during the exhaust stroke, the air is pushed out through another valve into the pressure tank.

Water vapor in the compressed air must be removed, or it will eventually damage the pneumatic components. Removing the moisture in the air is done by the **dryer**, of which there are several types. One type, an **aftercooler**, chills the air, causing the moisture to condense into drops, which can then be drained off. The **desiccant dryer** circulates the air through a moisture-absorbing chemical called a desiccant. When the desiccant becomes saturated (often indicated by a change in color), it must be changed.

The **receiver tank** (pressure tank) receives air from the compressor and becomes the high-pressure air reservoir for the system. Figure 10.25 shows the compressor and tank system. Power to the compressor motor is controlled by a pressure switch on the tank. When the tank pressure falls below a set value, the switch closes, and the compressor motor starts. When the tank pressure increases to a specified value, the switch opens, turning off the compressor. One benefit of using the tank is that it tends to smooth out the pulsations of air pressure that result from a reciprocating piston compressor. The typical pressure range for pneumatic systems is 30-150 psi (200-1050 kPa), although some systems go as high as 750 psi (5000 kPa).

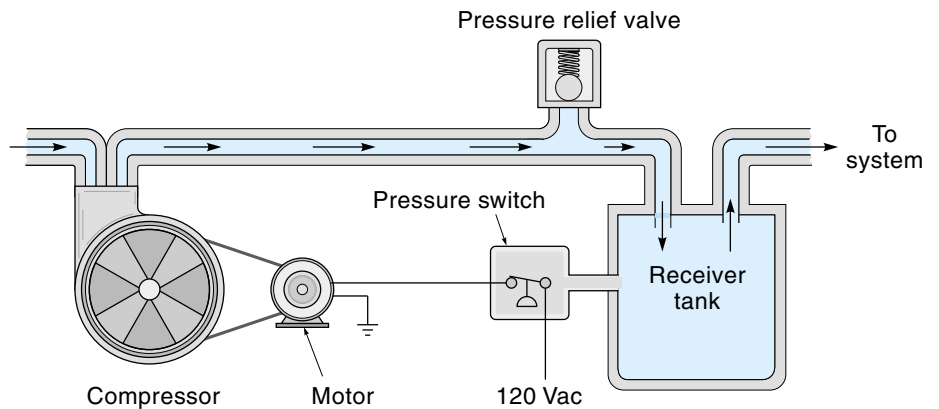
**Figure 10.24**  
A reciprocating piston compressor.



**Pressure Regulators**

The tank pressure can range anywhere between the high and low limits of the pressure switch. Some systems cannot tolerate this variation and hence require a pressure regulator to be installed between the tank and the system components. The **pressure regulator** can supply air at a constant pressure regardless of the source pressure *as long as the source pressure stays above the desired regulated pressure*.

**Figure 10.25**  
A receiver tank (pressure tank) system showing a pressure switch, which controls the compressor.



**Figure 10.26**  
A pneumatic  
pressure regulator.

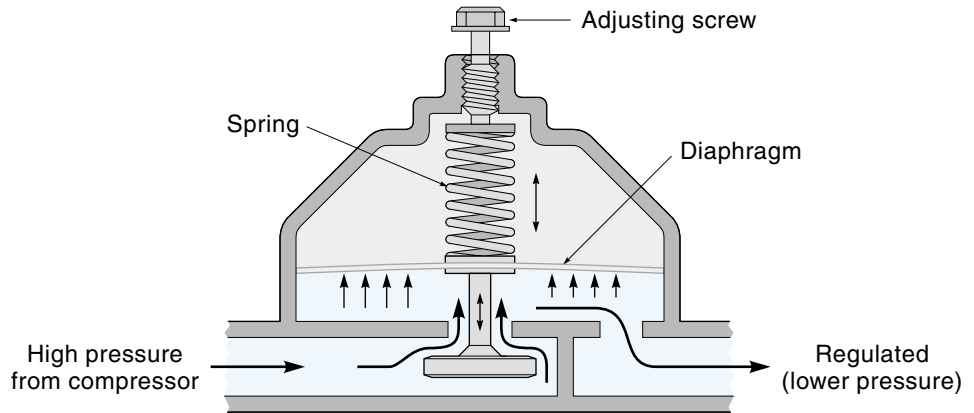


Figure 10.26 shows a simple pressure regulator. A spring-loaded diaphragm is pushed on by regulated air pressure. If the regulated air pressure starts to fall, the reduced pressure on the diaphragm causes it to move downward, thus opening a valve and allowing more high-pressure air in.

## Pneumatic Control Valves

**Pneumatic control valves** regulate the air flow, which in most cases means on or off. Many configurations are possible, but of particular interest is the bidirectional control valve that causes a piston to move in either direction. A pneumatic bidirectional control valve is shown in Figure 10.27. It is functionally similar to the hydraulic control valve shown in Figure 10.21, but there are some differences. For instance, pneumatic control valves use *O rings* (rubber seals) to minimize internal leaks. Pneumatic valves are usually designed to be full on or completely off and are driven by either an electric solenoid or a pneumatic control signal. Finally, these valves require only one tube coming from the air supply; the “used” air from the cylinder is simply vented to the atmosphere.

## Pneumatic Actuators

Pneumatic actuators convert air pressure into mechanical motion. There are two basic types: linear actuators (cylinder/piston or diaphragm types) and rotary actuators. Piston and rotary actuators are functionally similar to their hydraulic counterparts.

**Pneumatic cylinders** are available in a variety of shapes and sizes (Figure 10.29). As shown in Figure 10.28, there are two basic internal configurations. The double-acting cylinder connects to the valve with two tubes and can be driven in either direction. The **single-acting cylinder** can only be driven in one direction with air pressure and is returned by a spring.

**Figure 10.27**

Cutaway view of a pneumatic control valve. (Courtesy of Festo Corp.)

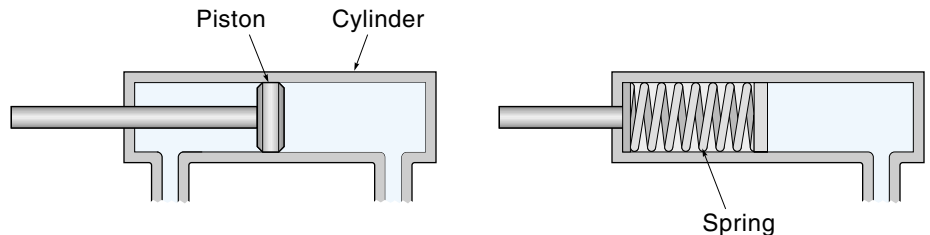


**EXAMPLE 10.5**

A spring-loaded single-acting cylinder has a diameter of 2 in. and a stroke of 2 in. The return spring has a spring constant of 3 lb/in. The available air pressure is 30 psi. What force can this cylinder supply to a load at the end of its stroke?

**Figure 10.28**

Two types of pneumatic cylinders.



(a) Double-acting cylinder

(b) Single-acting cylinder

**Figure 10.29**

Pneumatic cylinders.  
(Courtesy of  
Festo Corp.)



### SOLUTION

The total force that the piston can supply can be calculated from Equation 10.1 in exactly the same manner as it was done for the hydraulic cylinder. First, we calculate the piston surface area:

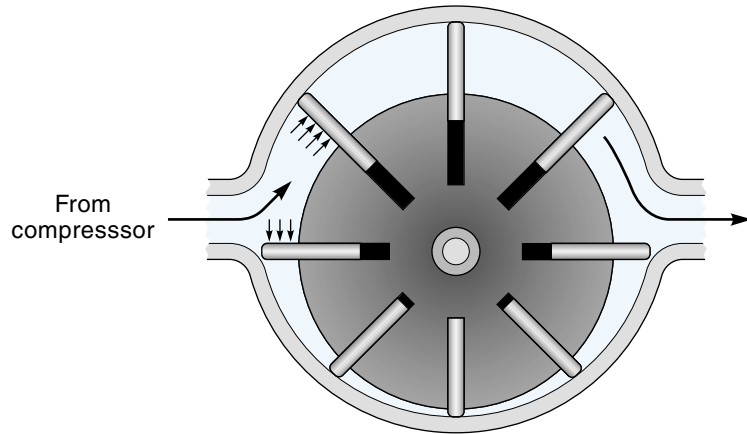
$$A = \pi r^2 = 3.14 \times 1 \text{ in}^2 = 3.14 \text{ in}^2$$

Now find the force from Equation 10.1:



**Figure 10.30**

A pneumatic vane motor.



$$F = PA = \frac{30 \text{ lb}}{\text{in}^2} \times 3.14 \text{ in}^2 = 94.2 \text{ lb}$$

Some of this force is used to compress the spring and is *not* available to the external load. The force needed to compress the spring 2 in. is calculated as follows:

$$F = \text{spring constant} \times \text{length} = \frac{3 \text{ lb}}{\text{in.}} \times 2 \text{ in.} = 6 \text{ lb}$$

Therefore, the total available force from the cylinder is

$$F = \text{piston force} - \text{spring force} = 94.2 \text{ lb} - 6 \text{ lb} = 88.2 \text{ lb}$$

Rotary actuators convert air pressure into rotary mechanical motion. One common design is the **vane motor** (Figure 10.30). The motor consists of a rotor that is offset in a housing. Protruding from the rotor are spring-loaded vanes that seal against the housing and slide in-and-out of the rotor as it turns. Motion is achieved because the vanes on the top have more exposed surface area than those on the bottom and hence receive more force, causing the rotor to turn clockwise.

## 10.4 FLOW-CONTROL VALVES

One common type of actuator used in process control systems is the **flow-control valve**, which regulates the flow of fluids. The control valve has a built-in valve-operating mechanism, allowing it to be controlled remotely by a signal from the controller. Usually, this signal is either electric or pneumatic.

**Figure 10.31**

Flow-control valves.

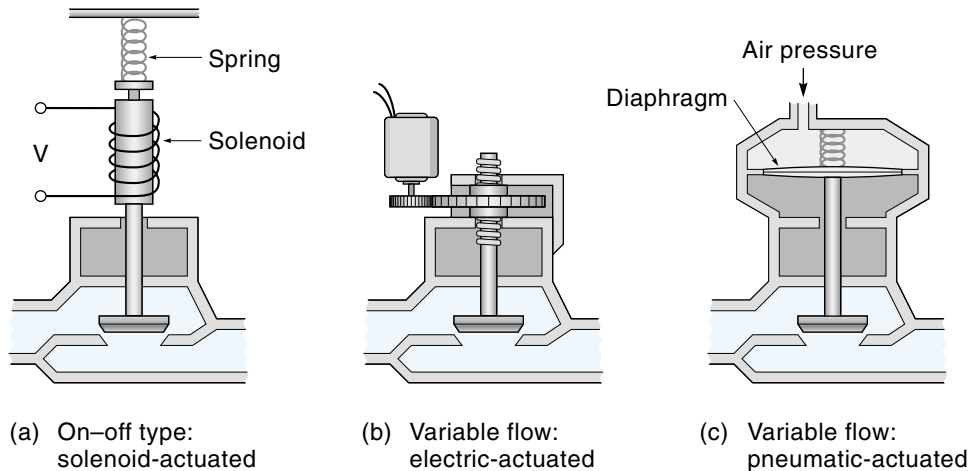
(a) On-off type:  
solenoid-actuated(b) Variable flow:  
electric-actuated(c) Variable flow:  
pneumatic-actuated

Figure 10.31 shows three types of flow-control valves. Figure 10.31(a) shows a solenoid-actuated, on-off valve. When the solenoid is energized, the valve is pulled open, and the fluid flows. When the solenoid is deenergized, a spring returns the valve to the closed position. On-off valves are used in batch processes (for example, a washing machine where the tank is filled to a specified level as quickly as possible, agitated for a while, then emptied).

Many processes require the ability to vary the flow of a fluid in a pipe on a continuous basis. To do this, the valve stem must be controlled with a linear actuator of some type. Figure 10.31(b) shows an electrically operated valve. In this case, an electric motor drives a leadscrew-type valve stem, so it can be put in any position.

Pneumatically operated valves use air pressure as the control signal. Shown in Figure 10.31(c), you can see that as the air pressure is increased, the diaphragm will move down (against a spring) and close the valve. This type of valve could be used in an on-off or a variable-flow application.

## SUMMARY

A linear actuator generates straight-line motion. There are three basic types of linear actuators: electric, hydraulic, and pneumatic.

Electric leadscrew actuators use the principle of a nut unscrewing along a threaded shaft. The threaded shaft (the lead screw) is rotated by an electric motor, and the nut moves linearly. The solenoid and electric linear motor use magnetics to create linear motion. Compared with hydraulic systems, electric linear actuators are clean and self-contained but, generally speaking, are not as strong and cannot move as fast or have other constraints that make them unsuitable for some applications.

Hydraulic systems are used to create linear motion, but it takes more hardware. A typical hydraulic system includes a pump, a pressure-control valve, a directional flow-

control valve, and a piston/cylinder-type actuator. The pump takes fluid at atmospheric pressure from a tank and causes it to flow under pressure. The pressure-control valve bleeds off some of the high-pressure fluid (back into the tank) in order to maintain a constant system pressure. The directional flow-control valve directs the fluid into one end or the other of the cylinder, causing the piston to move.

Pneumatic systems are similar to hydraulic systems except that air is used instead of fluid. Because fluid is incompressible and air is not, an air cylinder is more springy, and it is difficult to control its position precisely. For this reason, most pneumatic systems use mechanical stops to determine the length of travel.

Control valves are used to regulate the flow of fluids. The flow-control valve has a built-in valve-operating mechanism, allowing it to be controlled remotely by a signal from the controller.

---

## GLOSSARY

**accumulator** Used in a hydraulic system, a spring-loaded tank connected into the high-pressure side of the system; the accumulator can store some fluid to be used at peak-demand periods.

**actuator** The source of energy or motion in a system; typical actuators would be electric motors or hydraulic cylinders.

**aftercooler** Used in pneumatic systems, a device that removes moisture from the air supply by chilling and thereby condensing any water vapor.

**air compressor** Used in pneumatic systems, a device that creates the supply of high-pressure air.

**axial piston pump** A type of hydraulic pump that uses a number of small pistons to pump the fluid, this is a variable-displacement pump because the stroke of the pistons is determined by the adjustable angle of the swash plate.

**ball bearing screw** A ball bearing type of a leadscrew, recirculating balls roll between the threaded shaft and the nut to remove almost all friction.

**cogging** The jerky motion of electric linear motors, especially at low speeds.

**directional control valve** Used in hydraulic and pneumatic systems, it can direct the fluid into either end of the actuator cylinder, allowing the piston to be moved in either direction.

**desiccant dryer** Used in pneumatic systems, a device that dries the air supply by passing the air through a water-absorbing chemical.

**double-acting cylinder** A hydraulic or pneumatic cylinder and piston assembly in which the piston can be driven in either direction.

**dryer** Used in pneumatic systems, a dryer that removes the moisture from the air supply.

**extension rate** The rate at which a linear actuator can extend or retract; the term is usually applied to electric linear actuators.

**flow-control valve** Used to control the flow of fluids. A valve that has a built-in valve-operating mechanism, allowing it to be controlled remotely by a signal from the controller.

**gear motor** A simple hydraulic motor in which fluid pushes on the teeth of a set of constantly meshed gears (in a housing), causing them to rotate.

**gear pump** A simple positive-displacement hydraulic fluid pump that uses two constantly meshed gears (in a housing).

**hydraulic cylinder** A linear actuator powered by fluid under pressure, it consists of a cylinder, enclosed at both ends with a piston inside. The piston is moved by admitting the pressurized fluid into either end of the cylinder.

**hydraulic system** A system that uses pressurized fluid (oil) to power actuators such as hydraulic cylinders.

**hydrostatic pressure** The pressure that exists inside a closed hydraulic system; the significance is that the pressure is the same everywhere within the system (assuming no velocity effects).

**leadscrew linear actuator** A type of linear actuator that uses an electric motor to rotate a threaded shaft; the linear motion is created by a nut advancing or retreating along the threads.

**linear actuator** A prime mover that causes movement in a straight line, such as a hydraulic cylinder.

**linear motor** A motor that converts electric power directly into linear motion. Basically, it is a rotary brushless DC motor (BLDC) that has been rolled out flat.

**pneumatic control valve** Used in pneumatic systems, a valve that controls air flow to the actuators.

**pneumatic cylinder** A linear actuator powered by compressed air, it consists of a cylinder, enclosed at both ends with a piston inside. The piston is moved by admitting the compressed air into either end of the cylinder.

**pneumatic system** A system that uses compressed air to power actuators such as pneumatic cylinders.

**positive-displacement pump** A hydraulic pump that expels a fixed volume of fluid for each revolution of the pump shaft.

**pressure-control valve** Used in a hydraulic system, an adjustable spring-loaded valve that can maintain a set pressure in a system by bleeding off excess fluid.

**pressure regulator** Used in pneumatic systems, a regulator that maintains a specified pressure.

**prime mover** A component that creates mechanical motion from some other energy source; the source of the first motion in the system.

**receiver tank** Used in pneumatic systems, a reservoir for compressed air.

**reciprocating piston compressor** An air compressor that uses pistons and one-way valves to compress the air.

**single-acting cylinder** A spring-loaded pneumatic cylinder; air pressure drives the piston in one direction, and spring pressure returns it.

**solenoid** An electromechanical device that uses an electromagnet to produce short-stroke linear motion.

**spool** The moving part within the bidirectional control valve.

**stroke** The overall linear travel distance of a linear actuator.

**swash plate** Part of the axial piston pump; when the angle of the swash plate is changed, the flow from the pump is changed.

**vane pump** A hydraulic pump that uses an offset rotor with retractable vanes.

**variable-displacement pump** A hydraulic pump that can be adjusted to change the flow while keeping the speed of rotation constant.

**vane motor** A device that creates rotary motion from pneumatic or hydraulic pressure. The construction is similar to a vane pump.

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## EXERCISES

### Section 10.1

1. Describe the operating principles of a leadscrew linear actuator and give the advantages/disadvantages of an electric leadscrew actuator compared with a hydraulic linear actuator.
2. The characteristics of an electric linear actuator are given in Figure 10.4. If this actuator were lifting a 200-lb load, how long would it take to lift the load 6 in.?
3. The characteristics of an electric linear actuator are given in Figure 10.4. If this actuator were lifting a 800-lb load, how long would it take to lift the load 12 in.?
4. A 400-lb load needs to be lifted 7 in. in 5 s. Can the actuator described by Figure 10.4 do the job?
5. A solenoid application requires a stroke of 1 in. with at least 4 oz of force through the entire stroke. Select a solenoid that will do this job from the list given in Table 10.1. Coil voltage is to be 24 Vdc, intermittent duty.

6. A solenoid application requires a stroke of  $\frac{3}{4}$  inch with a starting force of at least 15 oz. Select a solenoid that will do this job from the list given in Table 10.1. Coil voltage is to be 120 Vac, continuous duty.

### Section 10.2

7. Calculate the force delivered by a 4-in. diameter hydraulic cylinder with a fluid pressure of 500 psi.
8. Calculate the force delivered by a 2-in. diameter hydraulic cylinder with a fluid pressure of 1000 psi.
9. Calculate the force delivered by a 10-cm diameter hydraulic cylinder with a fluid pressure of 3500 kPa.
10. Calculate the force delivered by a 6-cm diameter hydraulic cylinder with a fluid pressure of 7000 kPa.
11. A hydraulic system is operating with a pressure of 200 psi and is required to lift a load of 1400 lb. Find the cylinder diameter required (to the nearest  $\frac{1}{4}$  inch).
12. A hydraulic jack has the same diagram as Figure 10.12, where the small piston has a diameter of 0.5 in. and the large piston has a diameter of 3 in. Find the output force (from the large piston) when the input force is 10 lb.
13. A hydraulic jack has the same diagram as Figure 10.12, where the small piston has a diameter of 0.35 in. and the large piston has a diameter of 4 in. Find the output force (from the large piston) when the input force is 10 lb.
14. A hydraulic jack has the same diagram as Figure 10.12, where the small piston has a diameter of 1 cm and the large piston has a diameter of 8 cm. Find the output force (from the large piston) when the input force is 45 N.
15. Draw a diagram of a complete hydraulic system, from tank to actuator, and identify each system component.

### Section 10.3

16. Explain why a pneumatic system can be described as springy when compared with a hydraulic system.
17. Calculate the force delivered by a 1-in. diameter pneumatic cylinder with an air pressure of 100 psi.
18. A pneumatic system uses air pressure at 50 psi. What diameter cylinder would be required to provide a force of 75 lb?
19. A pneumatic system uses air pressure at 75 psi. What diameter cylinder would be required to provide a force of 50 lb?
20. A pneumatic system uses air pressure at 500 kPa. What diameter cylinder would be required to provide a force of 200 N?
21. A spring-loaded, single-action pneumatic cylinder has a diameter of 1.5 in. and a stroke of 4 in. The return spring has a spring constant of 3 lb/in. The available air pressure is 20 psi. What force will this cylinder exert at the end of its stroke?
22. Draw a diagram of a complete pneumatic system, from the intake filter to the actuator, and identify each of the system components.

# CHAPTER 11

## Feedback Control Principles

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### OBJECTIVES

After studying this chapter, you should be able to:

- Understand the terms and operation of a closed-loop control-system block diagram.
- Describe the basic operation of on-off control systems.
- Understand the concept and operation of a proportional control system (including bias) and calculate the error and controller output, given the system gain and inputs.
- Understand the concept of dead band and calculate the dead-band range for a proportional control system.
- Understand the concept and characteristics of integral control.
- Understand the concept and characteristics of derivative control.
- Understand the concepts and characteristics of PID control.
- Explain the circuit operation of an analog controller.
- Explain the principles of operation of a digital controller, including programming concepts and sample rate.
- Understand the concept of stability and interpret a Bode plot.
- Implement two methods of tuning a process control system.
- Explain the principles of operation and applications of fuzzy logic controllers.

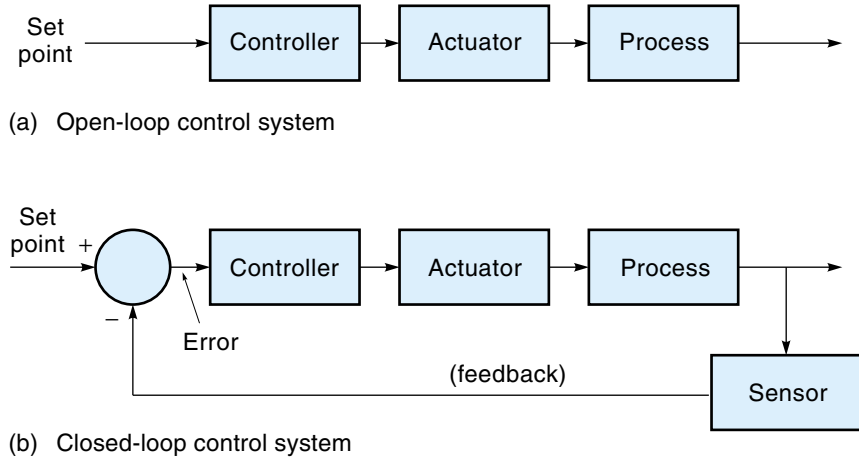
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### INTRODUCTION

Broadly speaking, control systems can be classified into two groups: open-loop and closed-loop. In an **open-loop system** [Figure 11.1(a)], no feedback is used, so the controller must independently determine what signal to send to the actuator. The trouble with this approach is that *the controller never actually knows if the actuator did what it was supposed to do.*

**Figure 11.1**

Open- and closed-loop systems.



In a **closed-loop system**, also known as a *feedback control system*, the output of the process is constantly monitored by a sensor [Figure 11.1(b)]. The sensor samples the system output and passes this information back to the controller. Because the controller knows what the system is actually doing, it can make any adjustments necessary to keep the output where it belongs. This self-correcting feature of closed-loop control makes it preferable over open-loop control in many applications. In this chapter, we deal with the principles and hardware of closed-loop controllers.

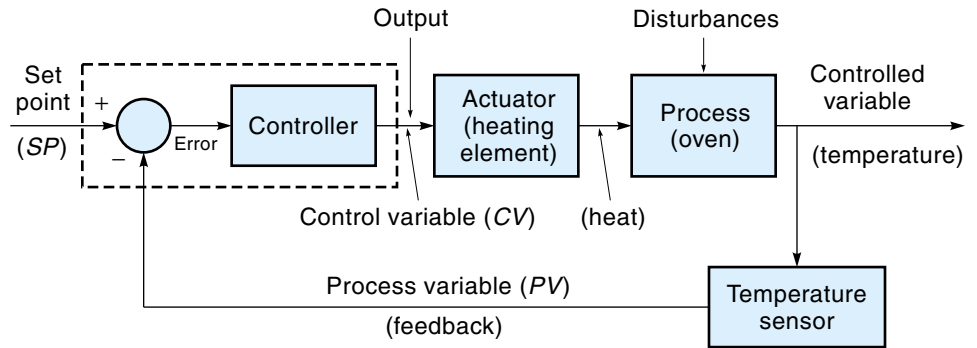
Clearly, the “heart” of the control system is the **controller**, an analog or digital circuit that accepts data from the sensors, makes a decision, and sends the appropriate commands to the actuator. In general, the controller is trying to keep the **controlled variable**—such as temperature, liquid level, position, or velocity—at a certain value called the **set point** (*SP*). A feedback control system does this by looking at the **error** (*E*) signal, which is the difference between where the controlled variable is and where it should be. Based on the error signal, the controller decides the magnitude and the direction of the signal to the actuator.

Figure 11.2 shows block diagrams of the two major classifications of feedback control systems: process control and servomechanisms.\* For a **process control system** [Figure 11.2(a)], the job of the controller is to maintain a stationary set point despite disturbances—for example, maintaining a constant temperature in an oven whether the door is opened or closed. Often, in a process control system, the set point is established by another controller acting in a supervisory role. In a **servomechanism** [Figure 11.2(b)], the job of the controller is to have the controlled variable track a moving set point—for example, moving a robot arm from one position to another.

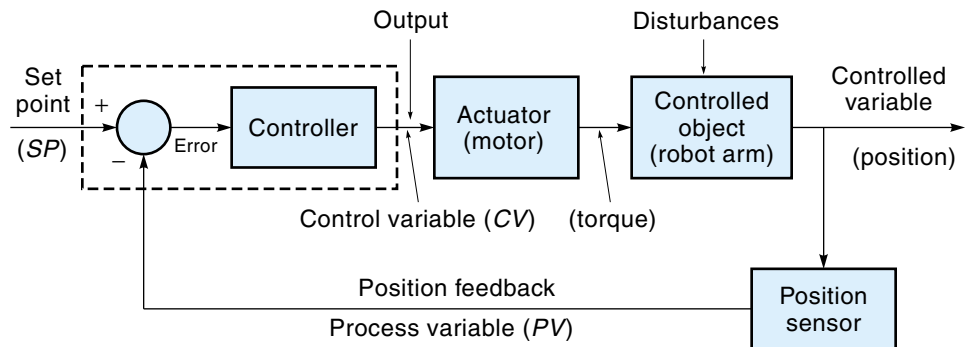
\*A word about terminology: Many different terms to describe the blocks and signals in Figure 11.2 are found in the literature. The terms selected for this text are a combination of the traditional and the modern.



**Figure 11.2**  
Feedback-control  
block diagrams.



(a) Process control system (heating system)



(b) Servomechanism (robot arm)

Engineers approach the problem of designing a controller for a servomechanism differently than they would for an industrial process control system. For a servomechanism such as a robot, the transfer functions of the individual system components are usually known or can be calculated. For example, the transfer functions of motors and sensors are usually provided by the manufacturer. The transfer function of mechanical components can be calculated based on the laws of physics (for example, moment of inertia) and empirical data. The response of the servomechanism can be modeled using higher mathematics, and the precise characteristics of the controller are then determined. On the other hand, for a large process control system, so many factors affect the system performance that engineers take a more empirical approach. This means that they use a general-purpose controller, and then tune it to meet the specifications of a particular system.

A new and increasingly important type of control is called fuzzy logic control. The fuzzy logic controller (Section 11.8) does not use mathematical models but mimics the skill and experience of a human operator.

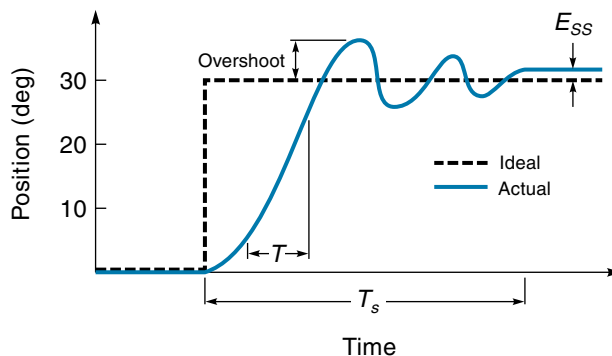
## 11.1 PERFORMANCE CRITERIA

Performance criteria are various measurable parameters that indicate how good (or bad) the control system is. These are divided into transient (moving) and steady-state (not changing) parameters.

The exact path the controlled variable takes when going from one position to the next is called its **transient response**. Consider the behavior of the robot arm whose response is shown in Figure 11.3; it is directed to move from 0 to 30°, as shown by the dashed line. This type of command (changing instantaneously from one position to another) is called a **step change**. The actual response of the system is shown as a solid line. As you can see, there is a difference between the ideal path of the arm and the one it took. One major consideration is how fast the system picks up speed (called **rise time**). The real arm simply cannot move fast enough to follow the ideal path. Rise time ( $T$ ) is usually defined as the time it takes for the controlled variable to go from 10 to 90% of the way to its new position. Another transient parameter is **overshoot**. Once the arm starts moving, its momentum will keep it going right on past where it was supposed to stop. Overshoot can be reduced by the controller but usually at the expense of a longer rise time. **Settling time** ( $T_s$ ) refers to the time it takes for the response to settle down to within some small percentage (typically 2-5%) of its final value. In this case, it is the time it takes for the oscillations to die out. Rise time, settling time, and overshoot are all related; a change in one will cause a change in the others.

The **steady-state error** ( $E_{SS}$ ) of the system is simply the final position error, which is the difference between where the controlled variable is and where it should be. In Figure 11.3,  $E_{SS}$  is shown as the position error after the oscillations have died out. This error is the result of friction, loading, and feedback-sensor accuracy. A sophisticated controller can reduce steady-state error to practically zero.

**Figure 11.3**  
Transient  
response.

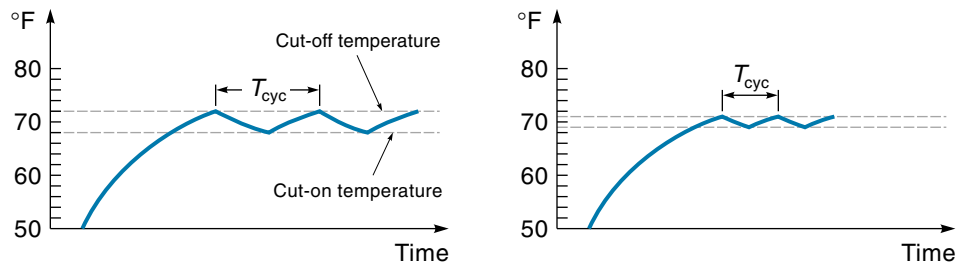


## 11.2 ON-OFF CONTROLLERS

### Two-Point Control

**Two-point control** (also called **on-off control**) is the simplest type of closed-loop control strategy. The actuator can push the controlled variable with only full force or no force. When the actuator is off, the controlled variable settles back to some rest state. A good example of two-point control is a thermostatically controlled heating system. Consider a house sitting for a long time with the heat turned off and an outside temperature of 50°F. Eventually, the inside temperature would drop to 50°. This is its rest-state temperature. Now suppose the heat is turned on and the thermostat is set for an average temperature of 70°. As Figure 11.4(a) shows, the inside temperature begins to climb, rapidly at first, and then more slowly (as the heat losses increase). When the temperature reaches the 72° cutoff point, the furnace shuts down. The house temperature immediately starts to decline toward its rest state of 50°; but long before it gets there, it reaches the cut-on point of 68°, and the furnace comes back on. Notice that the temperature curve is like a charging and discharging capacitor. A larger furnace would steepen the “charging” curve, and a larger house (or poorer insulation) would steepen the “discharging” curve (because the inside temperature would fall more quickly). Notice there is a cycle time ( $T_{cyc}$ ) associated with two-point control. This cycle time is affected by the capacity of the furnace and the house, as well as the temperature difference between the cut-on and cut-off points. If the limits were moved closer together—say, 69° and 71°, the temperature would be maintained closer to 70°, but the cycle frequency would increase, as illustrated in Figure 11.4(b). Generally, a high cycle rate is undesirable because of wear on motors and switches. Consequently, two-point control has only limited applications, mostly on slow-moving systems where it is acceptable for the controlled variable to move back-and-forth between the two limit points.

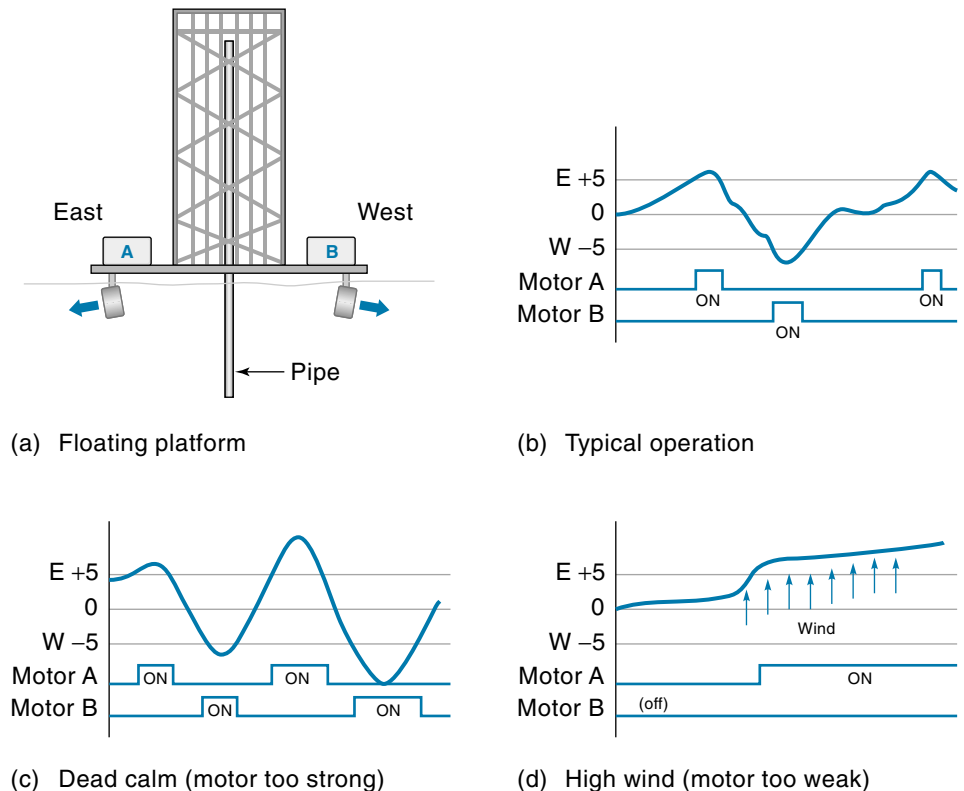
**Figure 11.4**  
Temperature curve of a two-point heating system.



### Three-Position Control

**Three-position control** is similar to two-point control, except in this case the controller has three states, such as forward–off–reverse, (or up–off–down, hot–off–cold, and so on). This strategy would be used in a system that has no particular self-seeking rest state. For example, consider the case of a floating oil-drilling platform that needs to stay over the wellhead on the ocean bottom, as illustrated in Figure 11.5(a). The platform must not drift more than 5 ft away from the center, or the pipe may break. Two motors, A and B, are used to keep the platform centered on the east-west axis (the north-south axis would be handled with other motors). If the platform drifts more than 5 ft east, motor A comes on and drives it back toward the center. Motor B will come on if the platform drifts more than 5 ft west. Figure 11.5(b) shows an example of the platform’s east-west movement. Notice that the controlled variable (platform position) will tend to oscillate back-and-forth across the center because the motor size selected was based on worst-case conditions (strong winds). In calm weather, a brief on-period of motor A may give the platform enough momentum to drift completely across the dead zone, only to be pushed back by motor B. In fact, if the system is not designed properly, the back-and-forth oscillations could get bigger and bigger, in which case the system has gone unsta-

**Figure 11.5**  
Example of a three-position control.



ble [Figure 11.5(c)]. On the other hand, if the wind is too strong for the motor, the platform will be driven outside the 5 ft limit, and the pipe will break [Figure 11.5(d)]. Three-position control is a simple control strategy that is appropriate for a limited number of applications, such as the one given.

## 11.3 PROPORTIONAL CONTROL

We now consider more sophisticated control strategies that require “smart” controllers that use op-amps or a microprocessor. The first and most basic of these strategies is called **proportional control**. With proportional control, the actuator *applies a corrective force that is proportional to the amount of error*, as expressed in Equation 11.1:

$$\text{Output}_p = K_p E \quad (11.1)$$

where

$\text{Output}_p$  = controller output due to proportional control (i.e., the corrective force)

$K_p$  = proportional constant for the system called **gain**

$E$  = error, the difference between where the controlled variable should be and where it is

Consider the position control system shown in Figure 11.6. A robot arm is powered by a motor/gearhead. A potentiometer provides position information, which is fed back to a comparator. This feedback signal is called the **process variable (PV)**. The comparator subtracts  $PV$  from the set point ( $SP$ ) to determine the error ( $E$ ) as expressed in Equation 11.2.

$$E = SP - PV \quad (11.2)$$

where

$E$  = error

$SP$  = set point, a desired value of the controlled variable

$PV$  = process variable, an actual value of the controlled variable

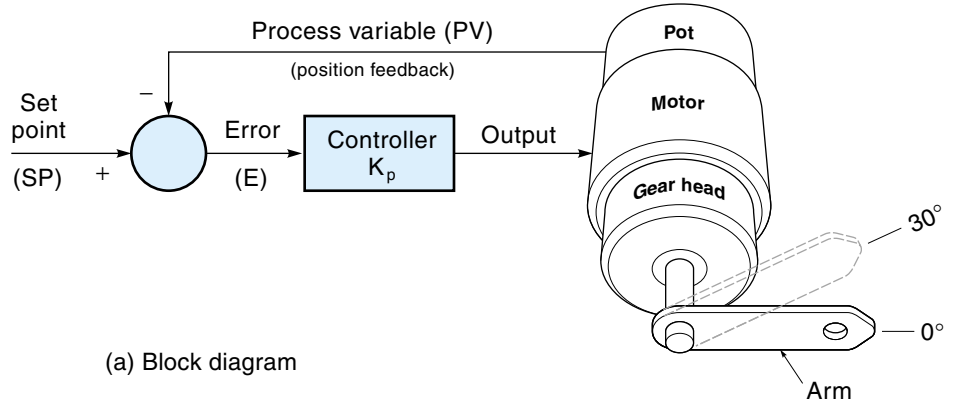
Referring to Equation 11.1, we see that the controller output is proportional to the error. This output directs the motor to move in a direction to reduce the error. As the position of the arm gets closer to the set point, the error diminishes, which causes the motor current to diminish. At some point, the error (and current) will get so small that the arm comes to a stop.

### EXAMPLE 11.1

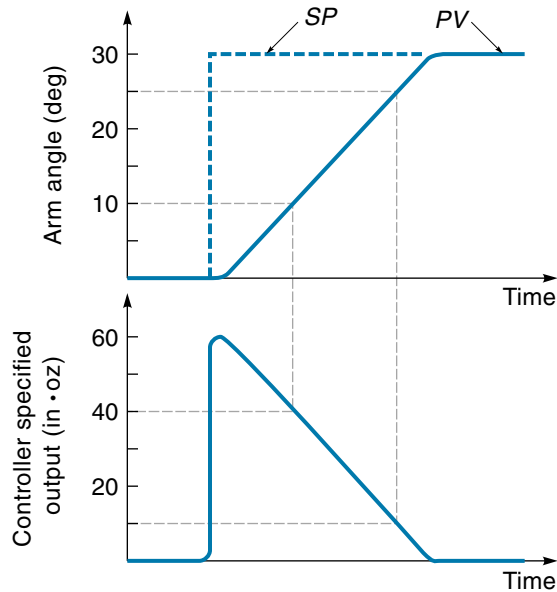
Assume that a motor driven arm was originally at  $0^\circ$  and then was directed to move to a new position at  $30^\circ$ . The gain of the system is  $K_p = 2 \text{ in.} \cdot \text{oz/deg}$ . Describe how the controller responds to this situation.

**Figure 11.6**

A proportional control position system.



(a) Block diagram



(b) Graphs showing response to change in set point

**SOLUTION**

The setup is shown in Figure 11.6. Originally, the arm is at rest at  $0^\circ$ . When the set point is first changed to  $30^\circ$ , an error signal of  $30^\circ$  results (because the arm is  $30^\circ$  away from where it should be):

$$\text{Error} = SP - PV = 30^\circ - 0^\circ = 30^\circ$$

Using Equation 11.1, we can calculate the initial restoring torque the system would generate:

$$\text{Output}_p = K_p E = 2 \text{ in.} \cdot \text{oz/deg} \times 30^\circ = 60 \text{ in.} \cdot \text{oz}$$

This means that the motor would initially be directed to create a torque of 60 in. · oz, causing the arm to rise rapidly. As the arm continues to rise, the controller output to the motor falls [Figure 11.6(b)].

When the arm gets up to 10°,

$$\text{Error} = SP - PV = 30^\circ - 10^\circ = 20^\circ$$

$$\text{Output}_p = K_p E = 2 \text{ in.} \cdot \text{oz/deg} \times 20^\circ = 40 \text{ in.} \cdot \text{oz}$$

The motor torque has now reduced to 40 in. · oz, so the arm will slow down.

When the arm gets up to 25°,

$$\text{Error} = SP - PV = 30^\circ - 25^\circ = 5^\circ$$

$$\text{Output}_p = K_p E = 2 \text{ in.} \cdot \text{oz/deg} \times 5^\circ = 10 \text{ in.} \cdot \text{oz}$$

With a motor torque of only 10 in. · oz, the arm is slowing way down.

Finally, when the arm reaches the new set point of 30°,

$$\text{Error} = SP - PV = 30^\circ - 30^\circ = 0^\circ$$

$$\text{Output}_p = K_p E = 2 \text{ in.} \cdot \text{oz/deg} \times 0^\circ = 0 \text{ in.} \cdot \text{oz}$$

With the motor torque at 0 in. · oz, the arm stops.

Having the correcting force be proportional to the error makes sense when you consider the following argument. A large error implies there is a long way to go, and so you want some speed to get there (which requires a large torque from the motor). However, when the error is small, the arm should slow down (small torque) so as not to overshoot. Also, note that the system is bidirectional in the sense that the torque will always be applied in whatever direction is needed to reduce the error. For example, consider again the system shown in Figure 11.6 (Example 11.1). When the arm was directed to go to 30°, the controller outputted to the motor a signal corresponding to 60 in. · oz of torque. Later, if the motor were directed to return to 0°, a new negative error would appear, causing a new (negative) motor torque to be generated:

$$\text{Error} = SP - PV = 0^\circ - 30^\circ = -30^\circ$$

$$\text{Output}_p = K_p E = 2 \text{ in.} \cdot \text{oz/deg} \times -30^\circ = -60 \text{ in.} \cdot \text{oz}$$

The negative sign of the output would result in a change in polarity of the applied voltage to the motor, which would cause it to run in the opposite direction. Thus, proportional control is capable of driving the arm in either direction.

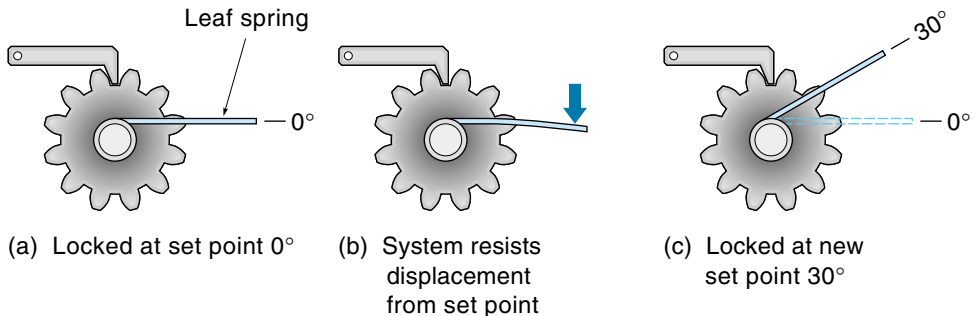
If you had the opportunity to experiment with the proportional control system shown in Figure 11.6, you would notice that the arm has a springy feel. For example, if you were to push down on the arm, you would feel a restoring force, as if you were pressing on a spring. The more you displaced it, the more it would resist. What you are actually feeling is the control system's correcting force, where the resistance to movement is proportional to the error (displacement). Figure 11.7 shows how we might model this property for the motor of Example 11.1. Notice in the model that the arm is a leaf spring attached to a locked shaft. If we try to push the arm down, the leaf spring resists. The farther the arm is deflected, the greater the resistance force [Figure 11.7 (b)]. When the external force is removed, the arm is returned by the spring to the center position. When we command the controller to move the arm to a new position (say,  $30^\circ$ ), it is like causing the entire spring to rotate, so it is now at rest in a new position and will resist being deflected from that point [Figure 11.7(c)].

### The Steady-State-Error Problem

Proportional control is simple, makes sense, and is the basis of most control systems, but it has one fundamental problem—steady-state error. In practical systems, proportional control cannot drive the controlled variable to zero error because as the load gets close to the desired position, the correcting force drops to near zero. This small force may not be enough to overcome friction, and the load comes to a stop just short of the mark.

Friction, *always present in mechanical systems*, is a nonlinear force that opposes the applied force. In fact, the load will not move at all until the friction force has been overcome. Figure 11.8(a) shows another spring model of a proportional control system. This time the load is sitting on a friction surface in the center of a frame to which it is connected with two springs. Figure 11.8(b) shows what happens when the frame quickly moves to the right. At first, the resulting push and pull from the springs drags the load to the right. As the load approaches the new center position, the declining force from

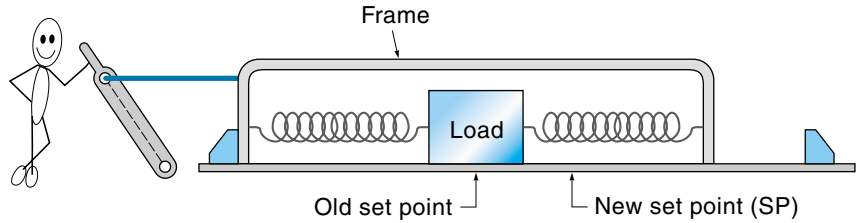
**Figure 11.7**  
Proportional control as modeled with a spring.



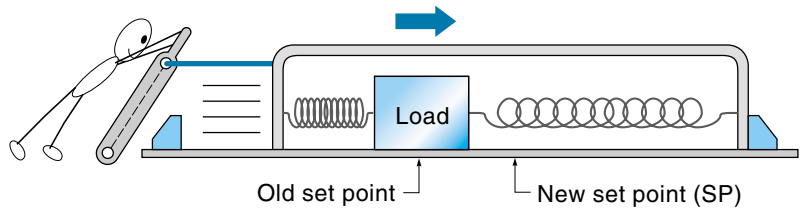


**Figure 11.8**

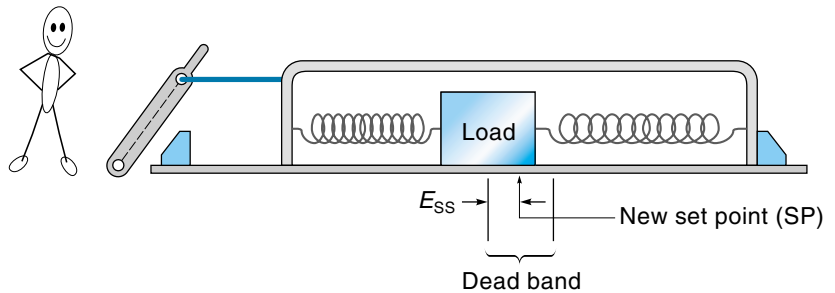
Friction causing steady-state error in a proportional system.



(a) Load is centered



(b) Load is commanded to new position by moving frame



(c) Load is at rest in new position; friction keeps load from being centered

the springs is overcome by friction, and the load stops *before* it gets to the new set point [Figure 11.7(c)]. The steady-state error is the distance between where the load stopped and the new set point. This region on either side of the set point, where the restoring force is incapable of precisely locating the controlled variable, is called the **dead band** or **dead zone**. Other factors (besides friction) also contribute to the dead band, those being, backlash, flexing of mechanical parts, and poor controller design.

**EXAMPLE 11.2**

A position control system has a gain  $K_p$  of 2 in. · oz/deg and works against a constant friction torque of 6 in. · oz. What is the size of the dead band?

### SOLUTION

To overcome the friction, the system must cause the motor to output 6 in. · oz. Because the input to the controller is the error signal, we need to find the value of error that results in a controller output of 6 in. · oz. Starting with the basic proportional equation (11.1),

$$\text{Output}_p = K_p E$$

Rearranging to solve for error,

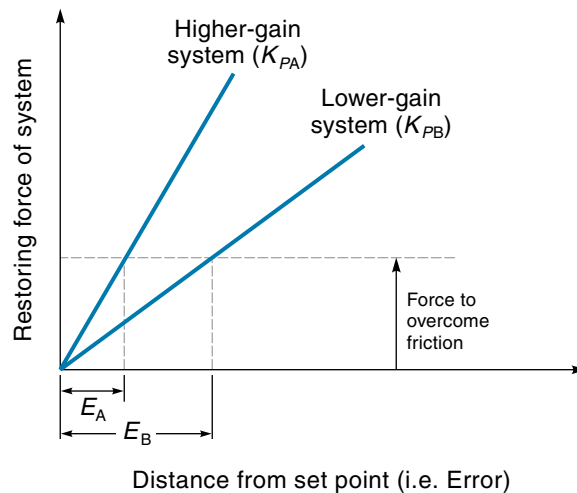
$$E = \frac{\text{output}_p}{K_p} = \frac{6 \text{ in} \cdot \text{oz}}{2 \text{ in} \cdot \text{oz/deg}} = 3^\circ$$

With  $3^\circ$  error on each side of the set point, dead band =  $6^\circ$ .

One way to decrease the steady-state error due to friction is to increase the system gain ( $K_p$  in Equation 11.1), which could be done in the model of Figure 11.8 by using stiffer springs. Consider the diagram in Figure 11.9. It shows the restoring force on the load for two different system gains,  $K_{pA}$  and  $K_{pB}$ . Notice in both cases that the restoring force is proportional to the error. The force necessary to overcome friction is shown as a dashed line. For the lower gain system, the error due to friction is shown as  $E_B$ . The higher gain system, acting like a stiffer spring, does not have to deflect as far from the set point to produce the force necessary to overcome friction. This system has less steady-state error, shown as  $E_A$ . It might seem reasonable to make the gain of every system very high; however, high  $K_p$  can lead to instability problems (oscillations). Increasing  $K_p$  independently without limit is *not* a sound control strategy.

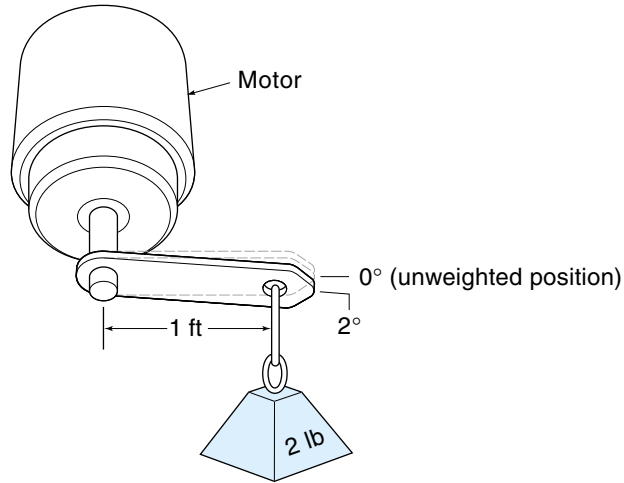
**Figure 11.9**

The higher-gain system has less steady-state error.



**Figure 11.10**

How gravity causes an arm to sag.



### The Gravity Problem

Another source of steady-state error is the *gravity problem*, which occurs when a constant external force is pushing on the controlled variable. Consider the robot arm lifting a weight in Figure 11.10. The set point for the arm is  $0^\circ$ , but when the weight is added, the arm sags. Its new position is where the restoring force of the system just balances the weight, in this case  $2 \text{ ft} \cdot \text{lb}$ . Herein is the paradox of the proportional control concept. *For the system to support the weight, there must be an error.* The reason? The proportional system only produces a restoring force when there is an error and the weight requires a constant force to support it. Increasing the system gain can reduce the error, but proportional control alone can never completely eliminate it.

#### EXAMPLE 11.3

Specify the system gain ( $K_p$ ) required to position the robot arm to within  $5^\circ$  of the set point (Figure 11.11). Total resistance from friction and gravity will be less than  $50 \text{ in} \cdot \text{oz}$ . The DC motor has a torque constant of  $25 \text{ in} \cdot \text{oz/A}$  with a  $10 : 1$  ratio gearbox built in. A  $350^\circ$  feedback pot is connected directly to the arm shaft.

#### SOLUTION

To keep the error down to  $5^\circ$ , the system needs to supply at least  $50 \text{ in} \cdot \text{oz}$  of restoring torque when the arm is  $5^\circ$  from the set point. This specifies the stiff-

ness, or gain ( $K_p$ ), of the system and is shown in graph form in Figure 11.12. To calculate the required value of  $K_p$ , start with Equation 11.1:

$$\text{Output}_p = K_p E$$

$$K_p = \frac{\text{output}_p}{E} = \frac{50 \text{ in.} \cdot \text{oz}}{5^\circ} = 10 \text{ in.} \cdot \text{oz/deg}$$

Thus, the controller must direct the motor to provide 10 in. · oz of torque for each degree that the arm is away from the set point. However, a controller cannot input degrees or output torque directly—the controller inputs and outputs voltages and currents.

The motor by itself provides 25 in. · oz/A of torque, but because it is geared down by a factor of 10, the output shaft torque is

$$10 \times 25 \text{ in.} \cdot \text{oz/A} = 250 \text{ in.} \cdot \text{oz/A}$$

which is the steady-state motor/gearhead transfer function. Finally, the pot is connected so that 10 V = 350°; thus,

$$\frac{10 \text{ V}}{350^\circ} = 0.029 \text{ V/deg}$$

which is the pot transfer function. To arrive at the gain required of the controller, multiply the system gain by the transfer functions:

$$K_p = \underbrace{\frac{10 \text{ in.} \cdot \text{oz}}{\text{deg}}}_{\text{Controller}} \times \underbrace{\frac{1 \text{ A}}{250 \text{ in.} \cdot \text{oz}}}_{K_p} \times \underbrace{\frac{1 \text{ deg}}{0.029 \text{ V}}}_{\text{Motor/gearhead}} = 1.38 \text{ A/V}$$

Controller
 $K_p$ 
Motor/gearhead
Pot

Thus, the controller must act like a power amplifier that supplies 1.38 A to the motor for each volt of input. For example, if we want to move the arm to the 30° position, we would input a voltage of

$$30^\circ \times 0.029 \text{ V/deg} = 0.87 \text{ V}$$

Assuming the arm had been at 0°, this command would cause an error signal of

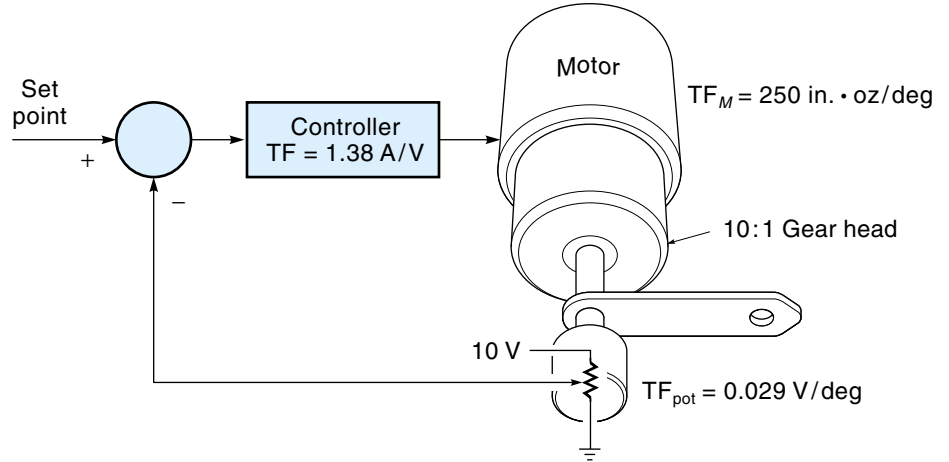
$$0.87 \text{ V} - 0 \text{ V} = 0.87 \text{ V}$$

The 0.87 V error signal would be converted by the amplifier to

$$0.87 \text{ V} \times 1.38 \text{ A/V} = 1.2 \text{ A}$$

**Figure 11.11**

The control system for Example 11.3.



This initial value of current would drive the motor/gearhead to produce a torque of

$$250 \text{ in.} \cdot \text{oz/A} \times 1.2 \text{ A} = 300 \text{ in.} \cdot \text{oz}$$

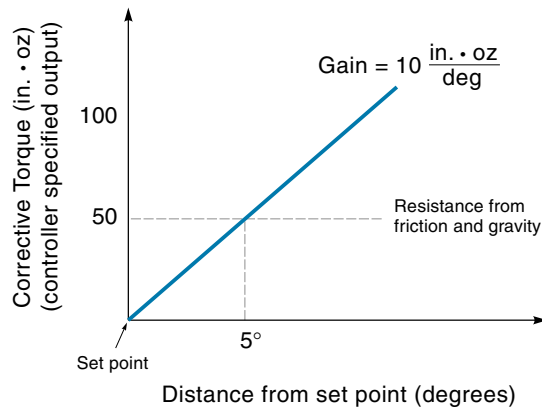
Eventually, as the error decreased, so would the torque. When the arm gets to 25° (5° short of 30°), the torque has dropped to 50 in. · oz, which (by design) is just enough to overcome friction.

**Bias**

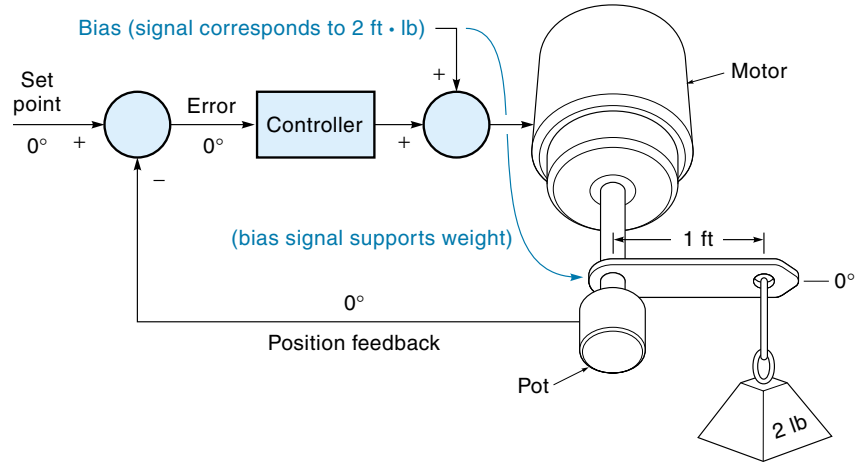
One way to deal with the gravity problem is to have the controller add in a constant value (to its output) that is just sufficient to support the weight. This value is called the

**Figure 11.12**

The stiffness of the control system in Example 11.3.



**Figure 11.13**  
Block diagram  
of control system  
with bias.



**bias** and is the same value that an open-loop system would give to the actuator. The equation for a proportional control system with bias is

$$\text{Output}_p = K_p E + \text{bias} \quad (11.3)$$

The bias value is considered to be another input to the controller, as shown in the block diagram of Figure 11.13. A 2-lb weight places a constant torque of 2 ft · lb on the motor. A bias signal (corresponding to 2 ft · lb) is added to the output of the controller and provides the drive signal to support the weight. This system allows the error to go to zero because the static load is being supported by the bias signal, not the proportional part of the controller. However, the bias must be changed if the static load is changed.

### Analog Proportional Controllers

An analog controller typically uses op-amps to provide the necessary gain and signal processing. For example, consider the flow control system illustrated in Figure 11.14(a). The controller's job is to maintain the flow of a liquid through a pipe at 6 gal/min. This system consists of (1) an electrically operated flow valve, (2) a flow sensor, and (3) the analog controller. The flow valve is operated with a signal of 0-5 V, where 0 V corresponds to completely closed and 5 V is all the way open. The flow sensor provides an output signal of 0-5 V, which correspond to 0-10 gal/min. The system is designed so that a sensor voltage swing of 2.5 V (50% of its range) will cause the flow valve to swing from full off to full on. Therefore, this system has what is called a 50% **proportional band**. This percent-type specification can be translated into a gain factor ( $K_p$ ) by rearranging Equation 11.1:

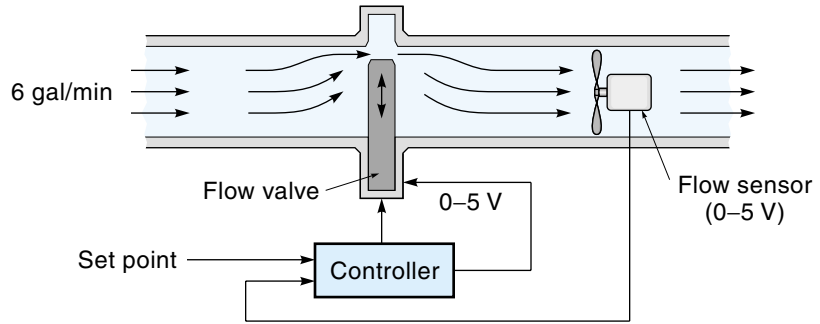
$$\text{Output}_p = K_p E$$

$$K_p = \frac{\text{output}_p}{E} = \frac{100\%}{50\%} = 2$$

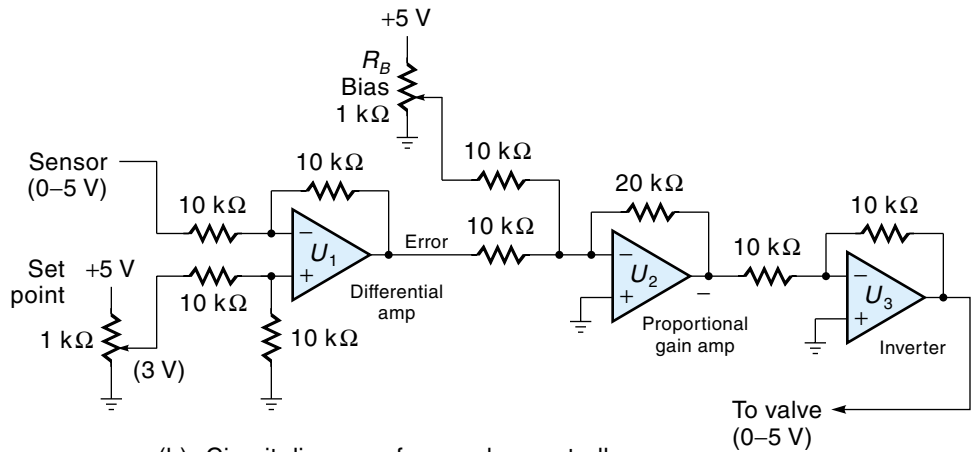
which is the proportional gain factor.

**Figure 11.14**

An analog controller for a proportional system.



(a) System diagram



(b) Circuit diagram of an analog controller

The analog controller circuit, shown in Figure 11.14(b), consists of three op-amps. The first op-amp ( $U_1$ ) is acting as a differential amplifier with a gain of 1, subtracting the sensor feedback signal from the set point to create the error voltage. To maintain a flow rate of 6 gal/min, the set point must be 3 Vdc as calculated below using the flow-sensor transfer function:

$$\text{Set point} = 6 \text{ gal/min} \times \underbrace{\frac{5 \text{ V}}{10 \text{ gal/min}}}_{\text{Flow-sensor}} = 3 \text{ Vdc}$$

The output of  $U_1$  (error signal) is fed into op-amp  $U_2$ , a simple (inverting) summing-type amplifier whose purpose is to provide the proportional gain ( $K_p$ ). To make the required gain of 2, the ratio of  $R_f/R_i$  ( $20 \text{ k}\Omega/10 \text{ k}\Omega$ ) is set to 2. Notice that the pot  $R_B$  can add a bias

voltage to the error signal if necessary. The output of  $U_2$  must be inverted to make the output positive; this is done with  $U_3$ , which is a simple inverting amplifier with unity gain.

## 11.4 INTEGRAL CONTROL

The introduction of **integral control** in a control system can reduce the steady-state error to zero. Integral control creates a restoring force that is proportional to the sum of all past errors multiplied by time, as expressed in Equation 11.4:

$$\text{Output}_I = K_I K_p \Sigma(E\Delta t) \quad (11.4)$$

where

Output<sub>I</sub> = controller output due to integral control

$K_I$  = integral gain constant (sometimes expressed as  $1/T_I$ )

$K_p$  = proportional gain constant

$\Sigma(E\Delta t)$  = sum of all past errors (multiplied by the time they existed)

For a constant value of error, the value of  $\Sigma(E\Delta t)$  will increase with time, causing the restoring force to get larger and larger. Eventually, the restoring force will get large enough to overcome friction and move the controlled variable in a direction to eliminate the error.

An analogy showing the power of integral control is a person who sits down in a comfortable chair to read a book. After a short time, the reader notices the dripping sound of a leaky faucet (steady-state error). The first response of the reader is to do nothing, but as time goes on the sink starts to fill up and spill over, which gets the reader's attention and he or she gets up and turns it off. The point is that the dripping (error) was not increasing, but the *effect* of the steady error was increasing with time until finally the reader (system) was motivated to do something about it.

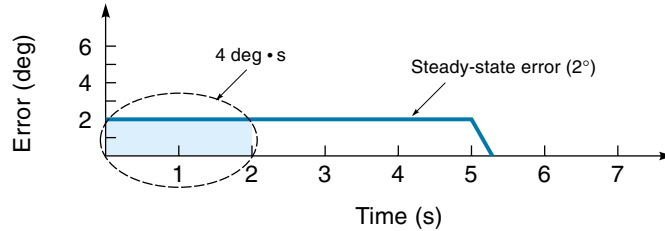
### EXAMPLE 11.4

Consider the case where a robot arm has a steady-state error position of  $2^\circ$  due to friction; this error is shown on the graph in Figure 11.15(a). As time elapses, the error remains at  $2^\circ$ . Figure 11.15(b) shows how the restoring torque due to integral control increases with time. The magnitude of the restoring torque at any point in time is proportional to the area under the error curve. For example, after 2 s the area under the error curve is  $4 \text{ deg} \cdot \text{s}$  (area =  $2 \text{ deg} \times 2 \text{ s}$ ), as illustrated in Figure 11.15(a). Assuming that  $K_p = 1 \text{ in} \cdot \text{oz/deg}$  and  $K_I = 1/\text{s}$ , then  $K_p K_I = 1 \text{ in} \cdot \text{oz/deg} \cdot \text{s}$ , and the restoring torque due to integral control alone is  $4 \text{ in} \cdot \text{oz}$  [Figure 11.15(b)]. After about 5 s,

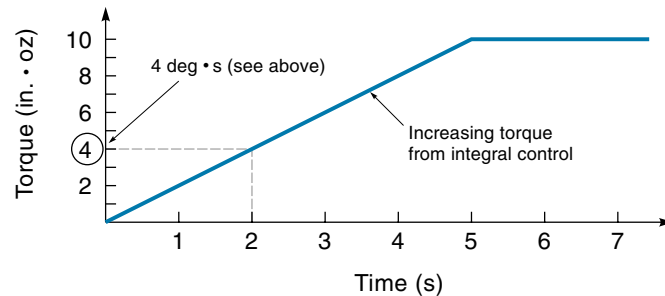


**Figure 11.15**

Graph showing integral control eliminating a steady-state error ( $K_{IKP}=1$ ).



(a) Steady-state error is being reduced to zero



(b) Output of integral controller

the restoring torque gets high enough to overcome friction and nudge the arm the last  $2^\circ$  to remove the error. Once the error goes to zero, the area under the curve stops growing, so the torque stops increasing; *however, it remains at the elevated level of 10 in. · oz*. This last point is important because it allows the gravity problem to be overcome as shown in Example 11.5.

### EXAMPLE 11.5

The proportional feedback system of Example 11.3 ( $K_p = 10$  in. · oz/deg) has been modified to include integral feedback. The arm has been at rest (at the  $30^\circ$  position) when a weight is placed on the end of the arm, causing a downward torque of 40 in. · oz [Figure 11.16(a)]. Describe how the control system responds to the weight.

### SOLUTION

The proportional gain of this system was determined to be 10 in. · oz/deg. As the following calculation shows, 40 in. · oz of torque from the weight would cause the arm to sag  $4^\circ$ :

$$\text{Output}_p = K_p E$$

$$E = \frac{\text{Output}_p}{K_p} = \frac{40 \text{ in.} \cdot \text{oz}}{10 \text{ in.} \cdot \text{oz/deg}} = 4^\circ \text{ sag}$$

which is shown at time A in Figure 11.16 (b).

The proportional control initially provides the entire 40 in. · oz of restoring force to support the arm, resulting in 4° of error. Because there *is* an error, however, integral control starts contributing to the restoring force. As time passes, the area under the error curve increases, and the integral control provides more and more of the 40 in. · oz needed to support the weight, while the proportional control provides less and less [shown as time B in Figure 11.16(b)]. Finally, at time C, the integral control is providing the entire 40 in. · oz, and the error is zero. A new steady state has been reached.

Integral control response can easily be observed on industrial robots. When a weight is placed on the arm, it will visibly sag and then restore itself to the original position.

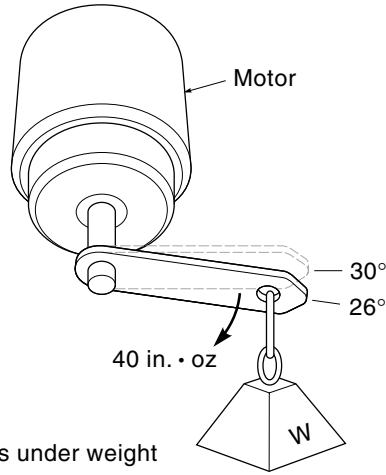
The integral constant  $K_I$  has traditionally been designated  $1/T_I$  and referred to as “repeats per minute” (or “repeats per second”). This is because (if  $K_I = 1$ ) the integral contribution will *increase* by the amount of the proportional contribution every time period. If  $K_I = 2$ , the integral contribution will increase by *double* the amount of the proportional contribution every time period, and so on. You can see this happening in Figure 11.15. Recall that  $K_p = 1^*$  and  $K_I = 1$  and that the steady-state error is 2 degrees [as shown in Figure 11.15(a)]. Therefore the contribution from proportional control alone would be 2 in · oz ( $K_p E = 1 \times 2 = 2$ ). Now look at Figure 11.15(b) and note that the integral contribution does indeed go up by the same amount of 2 in · oz each time period.

Although the addition of integral feedback eliminates the steady-state-error problem, it reduces the overall stability of the system. The problem occurs because integral feedback tends to make the system overshoot, which may lead to oscillations. One example of this problem is shown in the graphs of Figure 11.17. All mechanical systems have friction, and friction is nonlinear—that is, it takes more force to overcome friction when the object is at rest than it does to keep an object moving (a phenomenon often referred to as *sticktion*). At time = 0, the system in Figure 11.17 has just moved to a new position and stopped, leaving a steady-state error. The restoring force is equal to the contribution from proportional control plus the increasing force from the integral control. For a while the object doesn’t move, but finally the combined restoring force overcomes friction and the object “breaks loose.” Once moving, the friction force imme-

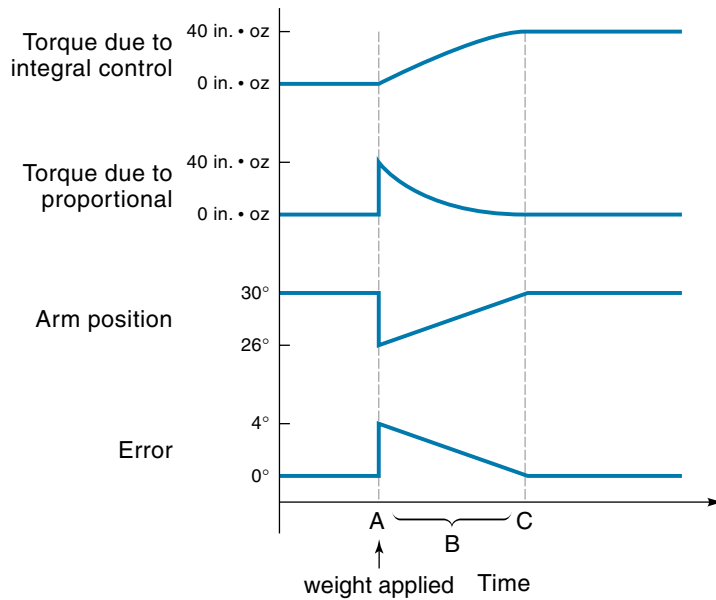
\*Units have been omitted for simplicity.

**Figure 11.16**

The system response of proportional plus integral control (Example 11.5).



(a) Arm sags under weight



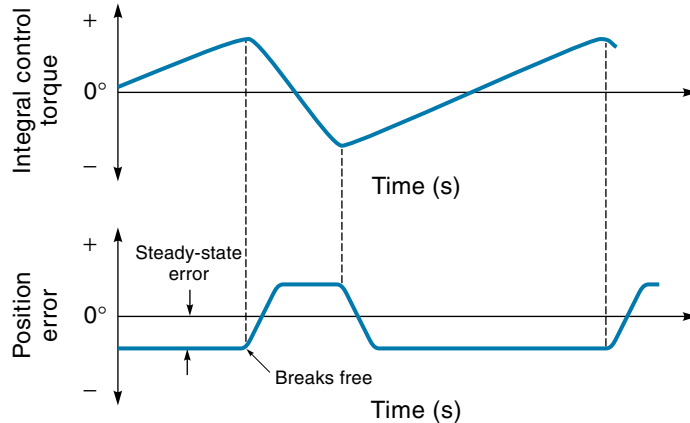
(b) Proportional and integral contributions

diately drops so some force is “left over,” which goes to accelerating the object. This may cause it to overshoot, and the whole process starts again from the other side.

The problem is that the proportional-integral system has no way (other than friction) to slow the object *before* it gets to the new set point. The system must overshoot before any active braking will be applied. So, unfortunately, the addition of integral feedback solves one problem, steady-state error, but it creates others: overshoot and

**Figure 11.17**

Integral control may cause overshoot and oscillations.



decreased stability. Also, the response of integral feedback is relatively slow because it takes a while for the error · time area to build up.

## 11.5 DERIVATIVE CONTROL

One solution to the overshoot problem is to include derivative control. **Derivative control** “applies the brakes,” slowing the controlled variable just before it reaches its destination. Mathematically, the contribution from derivative control is expressed in the following equation:

$$\text{Output}_D = K_D K_P \frac{\Delta E}{\Delta t} \quad (11.5)$$

where

Output<sub>D</sub> = controller output due to derivative control

$K_D$  = derivative gain constant (sometimes expressed  $T_D$ , unit is time)

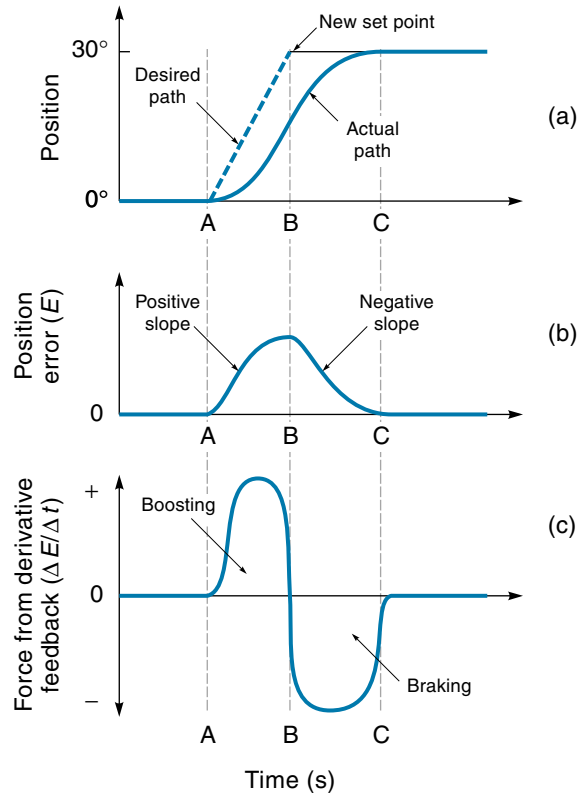
$K_P$  = proportional gain constant

$\frac{\Delta E}{\Delta t}$  = error rate of change (slope of error curve)

Figure 11.18 shows how a position control system with derivative feedback responds to a set-point change. Specifically, Figure 11.18(a) shows the actual and desired position of the controlled variable, Figure 11.18(b) shows the position error ( $E$ ), and Figure 11.18(c) shows the derivative control output. Assume the controlled variable is initially at  $0^\circ$ . Then at time  $A$  the set point moves rapidly to  $30^\circ$ . Because of mechanical inertia, it takes time for the object to get up to speed. Notice that the position error ( $E$ ) is increasing (positive slope) during this time period ( $A$  to  $B$ ). Therefore, derivative control, which is proportional to error slope, will have a positive output, which gives the object a boost, to help get it moving. As the controlled variable closes in on the set-point value ( $B$  to  $C$ ), the position error is decreasing (nega-

**Figure 11.18**

Contribution of derivative control, showing boosting and braking.



tive slope), so derivative feedback applies a negative force that acts like a brake, helping to slow the object.

For process control systems, where the set point is usually a fixed value, derivative control helps the system respond more quickly to load changes. For example, consider a controller that maintains a constant liquid level in a tank. If there is a rapid drop in liquid level (positive-slope error curve), derivative control responds by opening the inlet valve wider than proportional control alone would. Then, when the level is almost restored to the set point and the error is decreasing (negative-slope error curve), derivative control helps shut off the valve.

From the discussions so far, you can see that derivative control improves system performance in two ways. First, it provides an extra boost of force at the beginning of a change to promote faster action; second, it provides for braking when the object is closing in on the new set point. This braking action not only helps reduce overshoot but also tends to reduce steady-state error.

Naturally, the influence of derivative control on the system is proportional to  $K_D$  (in Equation 11.5); because derivative feedback improves overall system response, you might think that  $K_D$  should be as high as possible. However, like so many things in life,

too much of a good thing sometimes brings problems. In this case, too much derivative feedback will slow the system response and magnify any noise that may be present; what is needed is a balance between too much and too little. In the next section, we will describe methods to arrive at a numerical value for  $K_D$ . It is important to note that *derivative control has no influence on the accuracy of the system, just the response time, so it is never used by itself.*

## 11.6 PROPORTIONAL + INTEGRAL + DERIVATIVE CONTROL

Many control systems use a combination of the three types of feedback already discussed: **Proportional + Integral + Derivative (PID) Control**. The foundation of the system is proportional control. Adding integral control provides a means to eliminate steady-state error but may increase overshoot. Derivative control is good for getting sluggish systems moving faster and reduces the tendency to overshoot. The response of the PID system can be described by Equation 11.6, which simply adds together the three components required:

$$\text{Output}_{\text{PID}} = K_p E + K_I K_p \sum(E\Delta t) + K_D K_p \frac{\Delta E}{\Delta t} \quad (11.6)$$

Another form of this obtained by factoring out  $K_p$  from each term, is

$$\text{Output}_{\text{PID}} = K_p \left[ E + K_I \sum E\Delta t + K_D \frac{\Delta E}{\Delta t} \right] \quad (11.6a^*)$$

where

Output<sub>PID</sub> = output of the PID controller

$K_p$  = proportional control gain

$K_I$  = integral control gain (often seen as  $1/T_I$ )

$K_D$  = derivative control gain (often seen as  $T_D$ )

$E$  = error (deviation from the set point)

$\sum(E\Delta t)$  = sum of all past errors (area under the error · time curve)

$\Delta E/\Delta t$  = rate of change of error (slope of the error curve)

\*The traditional form of the PID equation is

$$\text{output}_{\text{PID}} = K_p \left[ E + \frac{1}{T_I} \int E dt + T_D \frac{dE}{dt} \right]$$

Comparing this with Equation 11.6a, you can see that the integral symbol ( $\int$ ) is another way of saying “the sum of” and the derivative  $dE/dt$  means “change in  $E$  divided by change in  $t$ .”

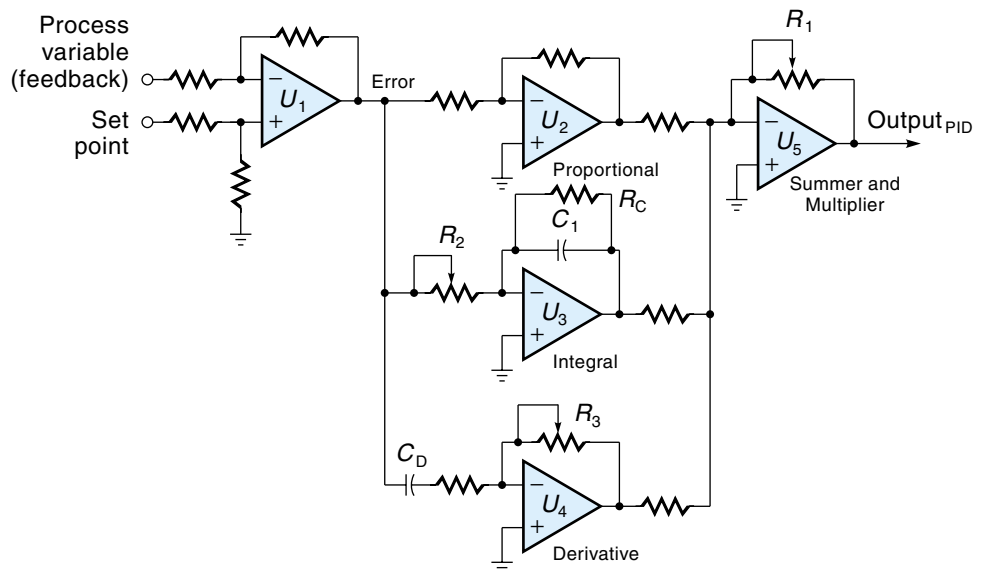
## Analog PID Controllers

The controller that implements the PID equation (11.6a) can be either analog or digital, but virtually all new installations are digital. Figure 11.19 shows a straightforward version of an analog PID controller that uses five differential amplifiers. The first differential amplifier ( $U_1$ ) subtracts the feedback from the set point to produce the error signal. Op-amps  $U_2$ ,  $U_3$ , and  $U_4$  are configured to be unit gain, integrator, and differentiator amplifiers, respectively. They produce the values for the three terms within the brackets in the PID equation 6.11(a). The final amplifier ( $U_5$ ) sums the three terms and multiplies the sum by  $K_p$  to produce the output. The capacitor  $C_I$  on the integrator accumulates the error in the form of charge, and the capacitor  $C_D$  of the differentiator passes only the *change* in error. The constants  $K_p$ ,  $K_I$ , and  $K_D$  are selected by adjusting  $R_1$ ,  $R_2$ , and  $R_3$ , respectively.

The circuit in Figure 11.19 can exactly implement the PID equation, but practical considerations usually complicate the picture. For example, all real amplifiers have an upper and lower limit at which points they become nonlinear. Large or rapidly changing error signals may cause either the integrator or differentiator amplifier to saturate. Should that happen, its output will temporarily dominate the output signal and may cause the system to go unstable.

One problem associated with the integrator is **windup**. This occurs when a system is subjected to a large disturbance, and the proportional controller (or actuator) in its attempt to correct the problem saturates “full on.” Because the system can’t provide as much output as is really needed, the error condition lasts longer than it theoretically

**Figure 11.19**  
An analog PID controller.



should, but all this time the integrator is accumulating. Consequently, when the error is finally reduced, the large accumulated integral factor may cause the controlled variable to overshoot. A solution to this problem is to have the integral control section disconnected when the system is saturated.

Another problem with integral control is that a true integrator will sum *all* the error · time area (since the beginning of time). It has been demonstrated that a better system results if the integrator slowly “forgets” the effects of errors from the distant past. This can be done by allowing the charge to slowly leak off capacitor  $C_I$  through  $R_C$  (in Figure 11.19).

A problem that can occur with derivative control is as follows: In a real control system, the set point is usually stepped up or down in discrete steps. A step change has an infinitely positive slope, which will saturate the derivative function. A solution to this problem is to base the derivative control on the feedback signal alone (*PV*) instead of the error because the controlled variable (be it temperature, position, or the like) can never actually change instantaneously, even if the set point does. The PID equation for this modified system is

$$\text{Output}_{\text{PID}} = K_p[E + K_I \sum(E\Delta t) + K_D \frac{\Delta PV}{\Delta t}] \quad (11.7^*)$$

where

Output<sub>PID</sub> = controller output

$K_p$ ,  $K_I$ ,  $K_D$  = gains for proportional, integral, and derivative

$E$  = error ( $SP - PV$ )

$PV$  = process variable (feedback from sensor)

A practical controller must account for various real-world problems such as those just discussed, but typically the nature and extent of design modifications required are not known until the actual system hardware is tested. Changing the characteristics of the analog controller may require component changes and/or rewiring. A distinct advantage of the digital controller, which we discuss next, is that its response is governed by software, which is relatively easy to change, even at the last minute.

## Digital PID Controllers

A PID digital controller is essentially a computer, most likely microprocessor-based. The controller executes a program that performs the same series of operations over and

\*The traditional form of Equation 11.7 is

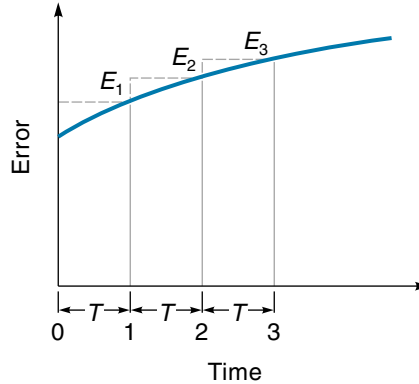
$$\text{Output}_{\text{PID}} = K_p \left[ E + \frac{1}{T_I} \int E dt + T_D \frac{dPV}{dt} \right]$$

where  $1/T_I = K_I$  and  $T_D = K_D$ .



**Figure 11.20**

A computer can approximate the area under the curve with rectangles.



over again. First, the computer inputs values of set point (*SP*) and process variable (*PV*). Then it uses these data to solve the system equation (11.6 or 11.7). Finally, it outputs the result to the actuator or the actuator drive circuit. One distinct advantage of the digital system is that control strategies and parameter constants can be changed or fine-tuned by simply modifying the software.

The following is a description of how the PID control equation (11.6a) could be implemented with a microprocessor-based controller. The first term inside the brackets of the PID equation is simply the error (*E*). Next comes the integral term  $K_I \sum(E\Delta t)$ . As has been discussed, this term represents the area under the error curve. Figure 11.20 shows a sample error versus time curve. The controller samples the error at regular intervals of time (*T*). A reasonable approximation of the total area under the curve is the sum of the rectangles, where the area of each rectangle is the product of *T* times the error *E*. This can be expressed in equation form as follows:

$$K_I \sum(E\Delta t) = K_I E_1 T + K_I E_2 T + K_I E_3 T \tag{11.8}$$

where

$K_I$  = integral gain

$E_1$  = error at time 1

$E_2$  = error at time 2, and so on

$T$  = time between the samples

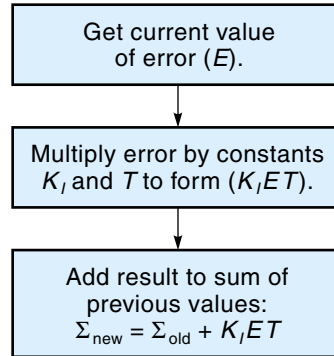
Equation 11.8 is called a *difference equation*, and it is very easy for a computer to evaluate. The program simply records the error value (*E*) at fixed intervals of *T* and keeps a running total of the  $K_I E T$  values. At any time, the integral term  $K_I \sum(E\Delta t)$  equals this total. Figure 11.21 shows the flowchart for this segment of the program.

The third term of the PID equation is the derivative component. The derivative, or rate of change, is actually the slope of the error curve. As shown in Figure 11.22, the slope is a function of the two most recent error values and the sample time (*T*):

$$\text{Slope} = \frac{\Delta E}{\Delta t} = \frac{(E_3 - E_2)}{T}$$

**Figure 11.21**

Flowchart to compute the integral term  $[K_I \sum(Et)]$ .



The derivative term in the PID equation (11.6a) is formed by multiplying the slope by the constant

$$K_D \frac{\Delta E}{\Delta T} = \frac{(K_D E_3 - K_D E_2)}{T} \tag{11.9}$$

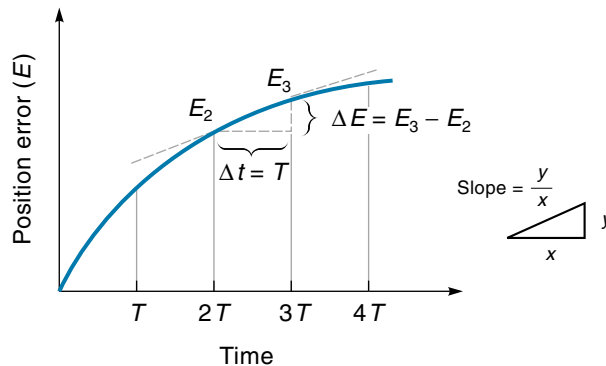
Equation 11.9 is another difference equation that is easily evaluated by the computer. The program would subtract the value of the older error term  $(K_D E_2)$  from the newer error term  $(K_D E_3)$  and then divide the difference by  $T$ . Figure 11.23 shows the flowchart for this segment of the program.

Having shown how the individual terms of the PID equation can be computed, we can now put them together and present Figure 11.24, a flowchart of the entire program. As you can see, the general format of the controller program is an endless loop. For each pass through the loop (called an **iteration** or **scan**), four basic operations are performed:

1. Read in the set point ( $SP$ ) and sensor data ( $PV$ ).
2. Calculate the individual components of error, integral, and derivative and then sum these.
3. Multiply the sum by  $K_p$  to form the PID output.
4. Send the PID output to the actuator.

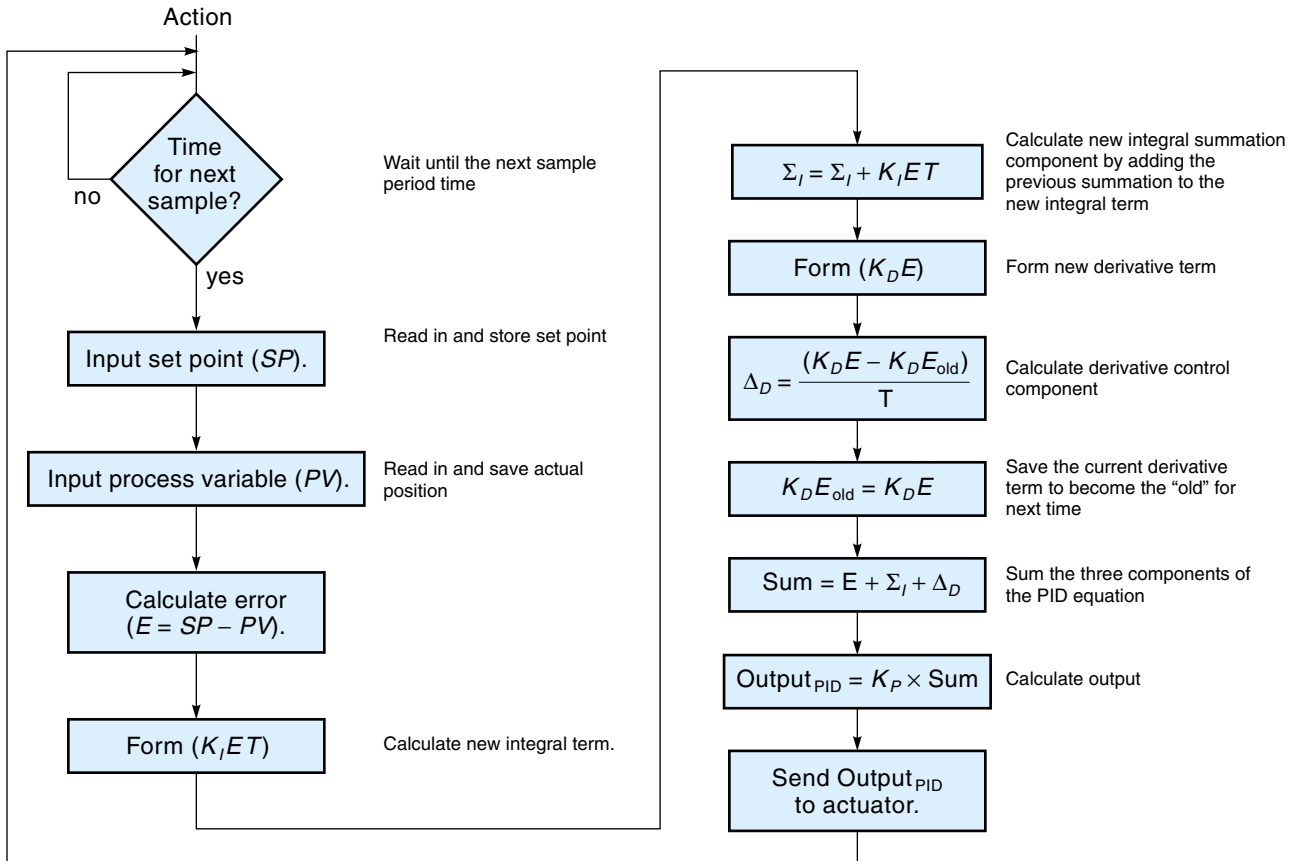
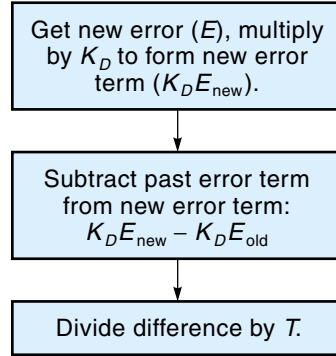
**Figure 11.22**

A computer can approximate  $\Delta E/\Delta t$  by finding the slope.



**Figure 11.23**

Flowchart to compute the derivative term ( $K_D \Delta E \Delta t$ ).

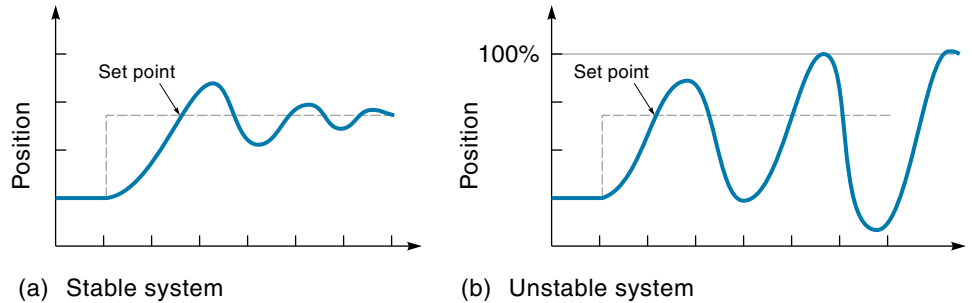


**Figure 11.24**

Flowchart for computing the PID equation.

**Figure 11.25**

Comparing a stable system and an unstable system.



### Stability

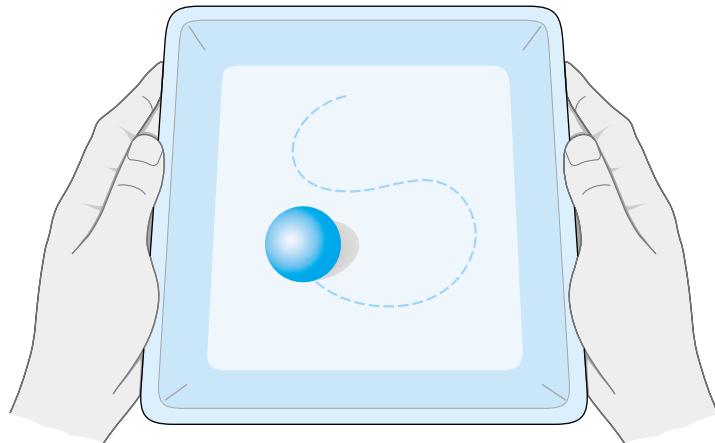
A stable system is one where the controlled variable will always settle out at or near the set point. An unstable system is one where, under some conditions, the controlled variable drifts away from the set point or breaks into oscillations that get larger and larger until the system saturates on each side. Figure 11.25 shows the response of stable and unstable systems. An unstable system is clearly not under control and may in fact be dangerous if large machines are involved.

There are many examples of the “oscillating” kind of instability in everyday life. Imagine holding a large wooden tray in your hands and trying to keep a golf ball (which is rolling about on the tray) centered at a certain spot (Figure 11.26). If the ball is on the right, you would lower the left side of the tray, but that causes the ball to roll past the center; now lower the right side of the tray to bring it back, and the ball rolls past center to the right, and so on. The ball ends up oscillating back-and-forth across the tray.

A primary reason control systems go into oscillation is because of phase lag caused by **dead time** or backlash, the interval between when the correcting signal is sent and when the system responds. Consider the case of driving at a moderate speed down a narrow lane. You would be making continuous, small back-and-forth adjustments to the steer-

**Figure 11.26**

A ball on a tray shows how control systems tend to oscillate.

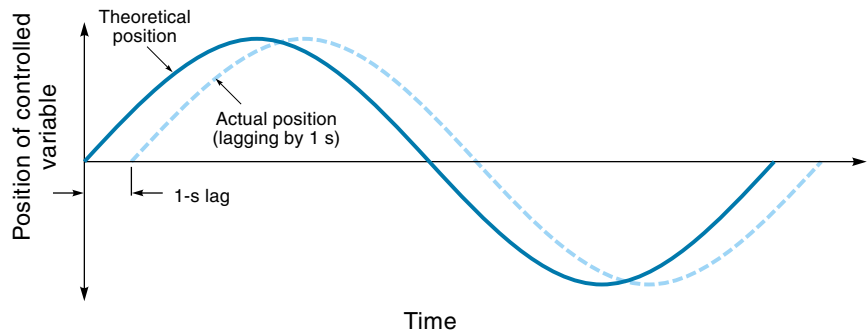


ing wheel in order to stay in the middle of the lane. Now imagine yourself driving down the same road but in a car with a lot of backlash in the steering wheel—in other words, there is half a turn of free play in the steering wheel before the car actually starts turning. If the car starts to drift toward the right, you turn the steering wheel toward the left; with all that free play, however, nothing happens initially, so you turn faster. Eventually, the slack is taken up, but by this time you are turning the steering wheel so fast that the car veers to the left, so you start frantically turning the wheel to the right, and so on.

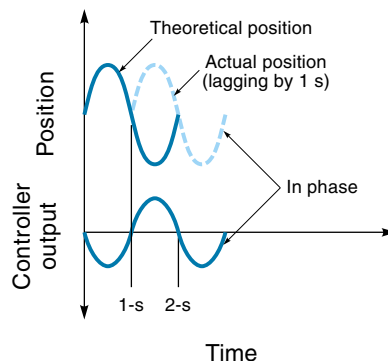
Closed-loop control systems employ negative feedback, which means that the controller is always pushing the system in the exact opposite direction of the error displacement. If the controlled variable happens to be oscillating, then the controlling force *should* be lagging by  $180^\circ$  because  $180^\circ$  is the opposite direction (for a rotational system). If there is any dead time in the system, the controller's response will lag even more than  $180^\circ$ . In other words, in an oscillating system, *lag time causes phase lag*, but the amount of phase lag depends on the frequency. For example, consider the case of a control system that is cycling at a frequency of 1 cycle per minute. If there is a 1-s lag between the controller output and the system response, then the delay is not very significant (as shown in Figure 11.27(a)).

**Figure 11.27**

How system lag can make the position of the controlled variable be in phase with the controller output.



(a) System with long cycle time; 1-s lag is not significant.



(b) System with 2-s cycle time; 1-s lag causes  $180^\circ$  lag.

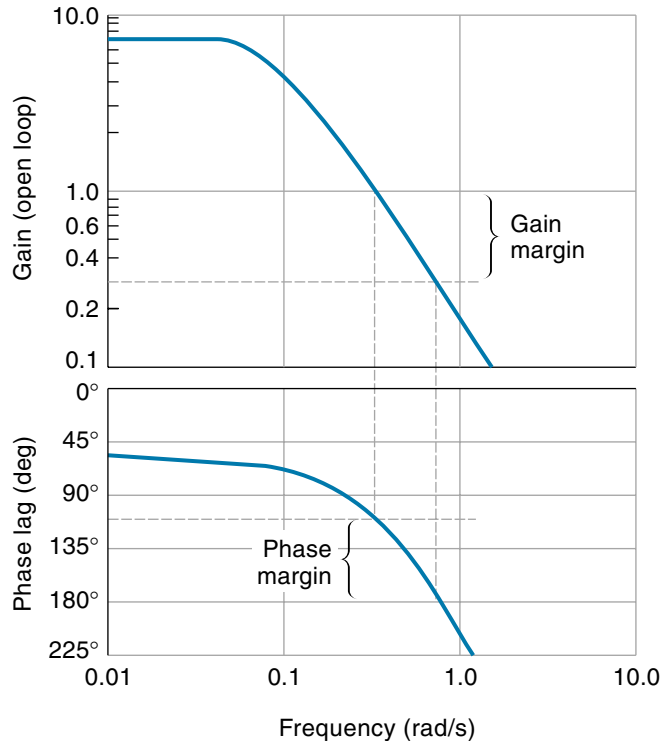
However, if the entire cycle time of the system is only 2 s, then a lag time of 1 s is very significant because the delay adds another  $180^\circ$  of phase lag, making a total lag of  $360^\circ$ . As Figure 11.27(b) shows,  $360^\circ$  lag is the same as being *in phase*, which means the system has positive feedback.

**Positive feedback** occurs when the controller output is in phase with the motion of the controlled variable so that, instead of applying a correcting force, the controller is simply helping the controlled variable go in any direction it wants. It's only a matter of time before some disturbance will cause the controlled variable to be pushed off center, and when it does, positive feedback will take over (if the system gain is at least 1) and push the controlled object to the edge of its range. Therefore, to guarantee that a system will be stable, *the gain must be less than 1 for any oscillating frequency where the lag causes an additional  $180^\circ$  of phase shift.*

A **Bode plot** is a graph that can help determine if a system is stable or not. As shown in Figure 11.28, the Bode plot is actually two curves sharing the same horizontal axis, which is frequency (in radians/second). The top graph shows how the open-loop gain varies with frequency, and the lower graph shows how the phase lag varies with frequency. The curves shown in Figure 11.28 are fairly typical in shape. We can see that the system represented by the Bode plot of Figure 11.28 is stable because, when the phase lag is  $180^\circ$ , the gain is less than 1 (0.3 to be exact). Two terms quantify the

**Figure 11.28**

A Bode plot showing stability considerations.



stability of a system: the gain margin and the phase margin. The **gain margin** is the gain safety margin (the difference between the actual gain and unity) taken at the frequency that causes  $180^\circ$  of phase shift. For the system shown in Figure 11.28, the gain margin is 0.7 ( $1 - 0.3 = 0.7$ ). The **phase margin** is the phase safety margin taken at the frequency that causes a gain of 1. For the system shown in Figure 11.28, the phase margin is  $70^\circ$  ( $180^\circ - 110^\circ = 70^\circ$ ).

### Tuning the PID Controller

The method of arriving at numerical values for the constants  $K_p$ ,  $K_I$  and  $K_D$  depends on the application. Traditionally, PID control was applied to process control systems. However, with the advent of small, fast, off-the-shelf PID modules, PID control is being applied to position control systems (such as robots) as well. In either case, a practical step-by-step procedure can be used to arrive at the PID constants. First, the constants  $K_p$ ,  $K_I$  and  $K_D$  are set to initial values, and the controller is connected to the system. The system could consist of the actual hardware or a computer simulation of same. Then the system is operated, and the response is observed. Based on the response, adjustments are made to  $K_p$ ,  $K_I$  and  $K_D$ , and the system is operated again. This iterative process of adjusting each constant in an orderly manner until the desired system response is achieved is called **tuning**. To make the system stable under all conditions, certain modifications may be necessary to the basic PID equation. Although many methods of tuning PID controllers exist, two of the most common were developed by Ziegler and Nichols and are called the continuous-cycle method and the reaction-curve method.

The **continuous-cycle method** (closed-loop method) can be used when harm isn't done if the system goes into oscillation. This method will yield a system with a quick response, which means a step-function input will cause a slight overshoot that settles out very quickly [Figure 11.29(b)]. The tuning procedure is as follows:

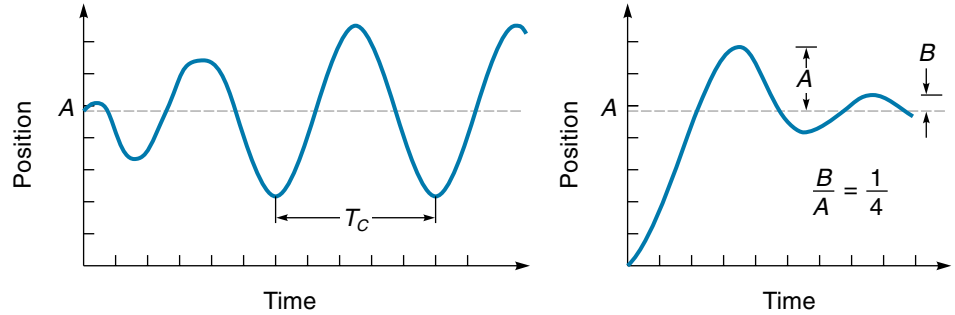
1. Set  $K_p = 1$ ,  $K_I = 0$ , and  $K_D = 0$  and connect the controller to the system.
2. Using manual control, adjust the system until it is operating in the middle of its range. Then increase the proportional gain ( $K'_p$ ) while forcing small disturbances to the set point (or the process) until the system oscillates with a constant amplitude, as shown in Figure 11.29(a). Record the  $K'_p$  and  $T_C$  for this condition.
3. Based on the values of  $K'_p$  and  $T_C$  from step 2, calculate the initial settings of  $K_p$ ,  $K_I^*$  and  $K_D^*$  as follows:

$$K_p = 0.6 K'_p \quad (11.10)$$

$$K_I = \frac{2}{T_C} \quad (11.11)$$

\* $K_I$  is often expressed as  $1/T_I$ , and  $K_D$  is sometimes expressed as  $I_D$ .

**Figure 11.29**  
Waveforms for the continuous-cycle method.



(a) System as forced into oscillation

(b) Resulting response after tuning

$$K_D = \frac{T_C}{8} \tag{11.12}$$

- Using the settings from step 3, operate the system, note the response, and make adjustments as called for. Increasing  $K_p$  will produce a stiffer and quicker response, increasing  $K_I$  will reduce the time it takes to settle out to zero error, and increasing  $K_D$  will decrease overshoot. Of course,  $K_p$ ,  $K_I$  and  $K_D$  do not act independently, so changing one constant will have an effect across the board on system response. Tuning the system is an iterative process of making smaller and smaller adjustments until the desired response is achieved [see Figure 11.29(b)].

### EXAMPLE 11.6

A control system is to be tuned using the continuous-cycle method. Initial settings were  $K_p = 1$ ,  $K_I = 0$ , and  $K_D = 0$ . By experiment it was found that the system first went into constant amplitude oscillations when  $K_p = 4$ . Figure 11.30 shows the system response during the experiment. Determine a first-cut set of values for  $K_p$ ,  $K_I$  and  $K_D$ .

### SOLUTION

From the response graph, we see that the period of oscillation is about 0.7 s. Therefore,  $T_C = 0.7$  s, and we can calculate the parameters  $K_p$ ,  $K_I$  and  $K_D$  using Equations 11.10, 11.11, and 11.12, respectively:

$$K_p = 0.6 K'_p = 0.6 \times 4 = 2.4$$

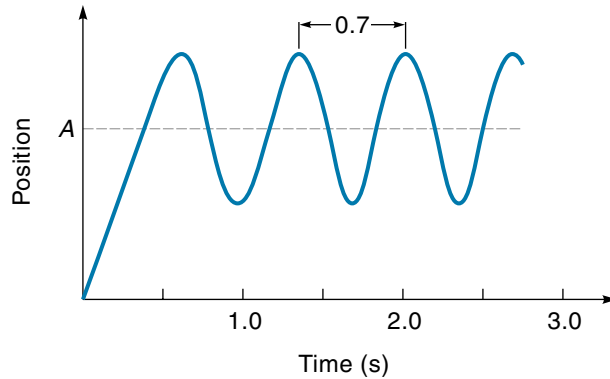
$$K_I = \frac{2}{T_C} = \frac{2}{0.7 \text{ s}} = 2.9/\text{s}$$

$$K_D = \frac{T_C}{8} = \frac{0.7 \text{ s}}{8} = 0.09 \text{ s}$$



**Figure 11.30**

Output of a test system  
(Example 11.6).

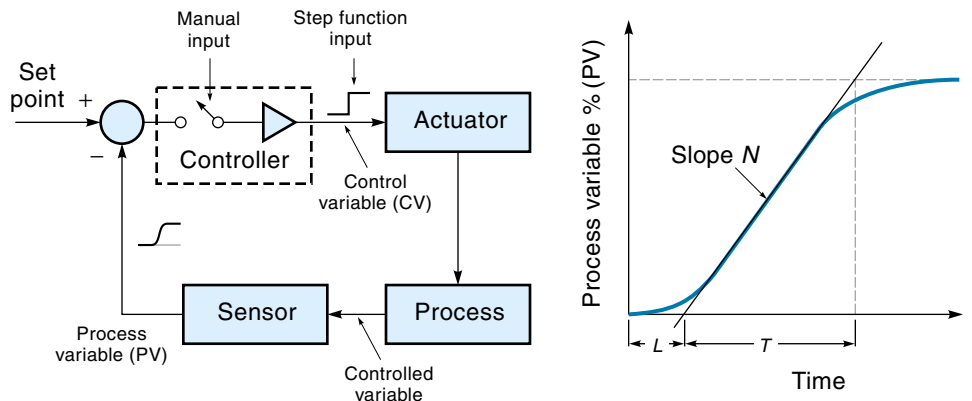


The **reaction-curve method** (open-loop method) is another way of determining initial settings of the PID parameters. This method does not require driving the system to oscillation. Instead, the feedback loop is opened, and the controller is manually directed to output a small step function to the actuator. The system response, as reported by the sensor, is used to calculate  $K_p$ ,  $K_I$  and  $K_D$ . Note that the actuator, the process itself, and the sensor are operational in this test, so their individual characteristics are accounted for. Because the loop is open, this procedure will work only for systems that are inherently stable.

One test possibility is shown in Figure 11.31(a). Here the loop was opened by placing the controller in the manual mode, then a small step function was manually introduced. This signal caused the controlled variable to move slightly, and the resulting position response was recorded. A typical response curve is shown in Figure 11.31(b). Note that the vertical axis corresponds to the range of the process variable (in percentage). The system constants are calculated based on the response curve, as outlined below:

**Figure 11.31**

Test setup and waveform for the reaction-curve method.



(a) Block diagram of test setup

(b) Graph of response to step input

1. Draw a line tangent to the rising part of the response curve. This line defines the lag time ( $L$ ) and rise time ( $T$ ) values. **Lag time** is the time delay between the controller output and the controlled variable's response.
2. Calculate the slope of the curve:

$$N = \frac{\Delta PV}{T} \quad (11.13)$$

where

$N$  = slope of the system-response curve

$\Delta PV$  = change in the process variable, as reported by the sensor (in percentage)

$T$  = rise time, from response curve

3. Calculate the PID constants :

$$K_p = \frac{1.2\Delta CV}{NL} \quad (11.14)$$

where

$\Delta CV$  = percent step change the in control variable (output of controller)

$N$  = slope, as determined by Equation 11.13

$L$  = lag time [see Figure 11.31(b)]

$$K_I = \frac{1}{2L} \quad (11.15)$$

$$K_D = 0.5 L \quad (11.16)$$

### EXAMPLE 11.7

The reaction-curve method is used to tune a PID control system. The system was turned on and allowed to reach a steady-state middle position. Then the controller was placed in the manual mode and directed to output a small step-function signal. The system response (from the sensor) was recorded. Figure 11.32 shows the step signal from the controller ( $CV$ ) and the resulting response ( $PV$ ).

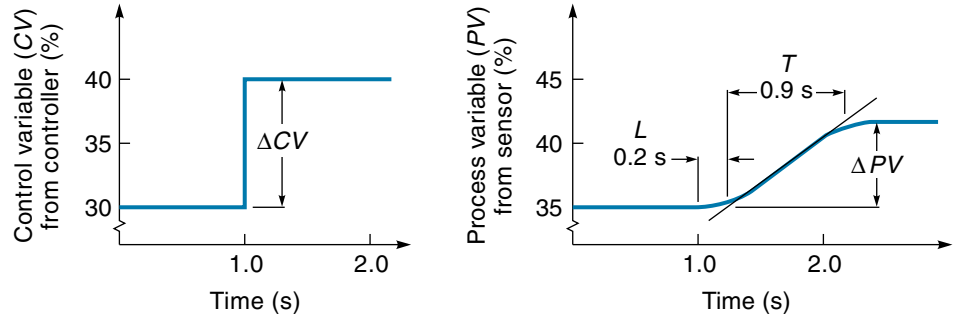
### SOLUTION

From Figure 11.32(a), we see that the signal from the controller ( $CV$ ) was a step of 10% ( $40\% - 30\% = 10\%$ ). Figure 11.32(b) shows the system response ( $PV$ ). The  $PV$  went from 35 to 42% for a change of 7%. After constructing the tangent line on Figure 11.32(b), we read from the graph the values of  $L$  and  $T$ :

$$L = 0.2 \text{ s}$$

$$T = 0.9 \text{ s}$$

**Figure 11.32**

 Waveform for  
 Example 11.7.


(a) Step input

(b) Response to step input

Next we calculate the slope of the response from Equation 11.13:

$$N = \frac{\Delta PV}{T} = \frac{7\%}{0.9 \text{ s}} = 7.8\%/s$$

Now we can calculate the PID constants by applying Equations 11.14-11.16, respectively:

$$K_p = \frac{1.2\Delta CV}{NL} = \frac{1.2 \times 10\%}{7.8\%/s \times 0.2 \text{ s}} = 7.7$$

$$K_I = \frac{1}{2L} = \frac{1}{2 \times 0.2 \text{ s}} = 2.5/s$$

$$K_D = 0.5L = 0.5 \times 2 \text{ s} = 0.1 \text{ s}$$

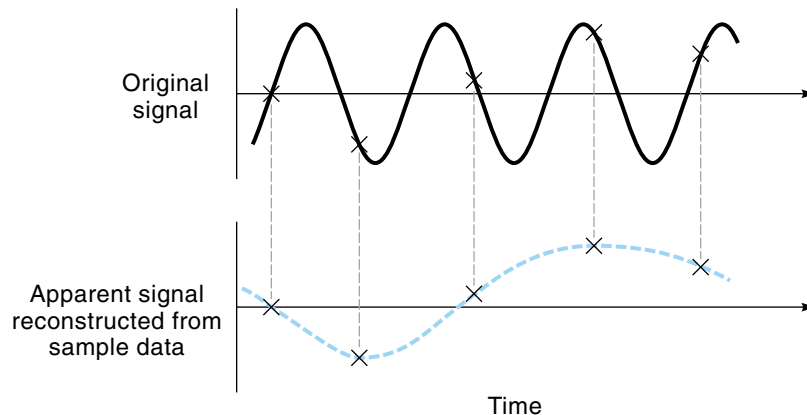
These values would be used as initial settings; the final settings would come from fine-tuning, as discussed earlier.

## Sampling Rate

In a digital control system, **sample rate** is the number of times per second a controller reads in sensor data and produces a new output value. Generally speaking, the slower the sample rate, the less responsive the system is going to be because the controller would be always working with “old” data—data that were current when the last sample was taken. Another problem with a slow sample rate is that the controller may simply not be able to keep up with what’s going on. Shannon’s sampling theorem states that the sampling rate must be at least twice the highest frequency being monitored or else **aliasing** can occur, when the collection of sampled data is not sufficient to re-create the original signal. Figure 11.33 shows this condition where the sample rate was slightly

**Figure 11.33**

Waveforms showing aliasing.



less than twice the frequency. Notice that the sample values would suggest a frequency one-third of what the actual frequency was. Of course, in a control system, the controlled variable is not usually tracing out a sine wave; it is more likely to be some random nonlinear motion. How would Shannon's sampling theorem be applied in such a case? To answer this question, we must invoke Fourier's theorem: *Any* function, sinusoidal or not, can be expressed as the sum of sine waves. It turns out that the more non-sinusoidal a signal becomes (say, approaching a square wave), the more significant higher frequencies it contains. This means that, to accurately follow a random path, a control system would have to sample at least twice as fast as the highest component frequency. In practice, a sampling rate of at least ten times the highest (visually apparent) frequency in the system is usually sufficient.

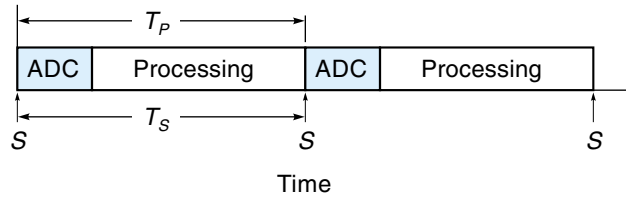
In most systems, sampling is done once at the beginning of each pass through the program loop (that is, one sample for each iteration). Therefore, the sampling rate is determined by how long it takes for the loop to be executed. Also, certain hardware functions take time, like an analog-to-digital conversion (typically, 100  $\mu$ s or less for 8 bits). Figure 11.34(a) illustrates the sampling cycle; Figure 11.34(b) shows one approach to speed up the sampling rate. In this case, the analog-to-digital conversion occurs at the same time as processing occurs. Although this increases the sample rate, it does not reduce the *throughput time*, the time for any individual sample to pass through the system.

### EXAMPLE 11.8

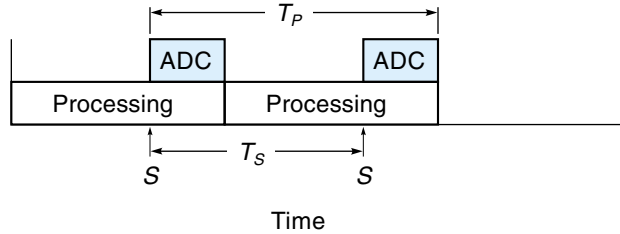
A microprocessor-based control system runs at a clock speed of 1 MHz. The system uses an 8-bit ADC with a 100- $\mu$ s conversion time. The program loop that processes the analog-to-digital input requires 55 instructions with an average execution time of 4 clocks/instruction. If analog-to-digital conversion is not overlapped with processing, what is the maximum sample rate? What is the highest frequency that this system can monitor?

**Figure 11.34**

Data-sampling cycles.



(a) ADC takes time



(b) Overlapping ADC saves time

$T_P$  = Time for one sample to be processed  
 $T_S$  = Time between samples  
 $S$  = Sample taken

### SOLUTION

Analog-to-digital conversion time =  $100 \mu\text{s}$

$$\begin{aligned} \text{Processing time} &= (55 \text{ instructions}) \times \frac{(4 \text{ clocks})(1 \mu\text{s})}{\text{instruction}} \\ &= 220 \mu\text{s} \end{aligned}$$

$$\text{Sample time} = 100 \mu\text{s} + 220 \mu\text{s} = 320 \mu\text{s}$$

$$\text{Maximum sample rate} = \frac{1}{320 \mu\text{s}} = 3.125 \text{ kHz}$$

With a sample rate of 3.125 kHz, this system should be able to track a controlled variable with a periodic motion of 312.5 kHz ( $3.125 \text{ kHz}/10 = 312.5 \text{ Hz}$ ).

### Autotuning

**Autotuning** is the capability of some digital controllers to monitor their own output and make minor changes in the gain constants ( $K_P$ ,  $K_I$  and  $K_D$ ). For example, a temperature-control system might decrease its proportional gain slightly if the overshoot is beyond a certain threshold and increase the gain if the response is too slow. Like manual tuning, autotuning is an iterative process, but because it is ongoing, the system can adapt to changes in the process. For this reason, controllers that use autotuning are known as **adaptive controllers**.

## 11.7 PIP CONTROLLERS

The set point has been defined as the place where you want the controlled variable to be. In a dynamic system, such as a robot arm, the desired position is a moving target, in which case we are concerned with path control. Further, the desired path between two points may not be a straight line. For example, a welding robot needs to follow the path of the seam. There are two ways to implement path control: the “carrot-and-horse” approach and the **feedforward**, or PIP approach.

The servomechanisms discussed so far have all used the carrot-and-horse idea: The controlled variable (horse) is always trying to catch up to the moving set point (carrot). The controller has information about only the past and the present, not where it is going, which is a severe handicap to place on the system. It is like a race-car driver, speeding around the track, who can only see one foot in front of the car. Normally, the driver can see in advance if the road changes direction and make the necessary adjustments to cause a smooth turn. Without the ability to look ahead, the driver would have to go very slow or risk driving off the road (overshoot) on a sharp turn.

A **Proportional + Integral + Preview (PIP)** controller is a system that incorporates information of the future path in its current output. Many systems have this information available—either the entire path is stored in memory or the system is equipped with a preview sensor as illustrated in Figure 11.35 for a welding robot. The equation for a simple PIP system follows:

$$\text{Output} = K_p E + K_i K_p \sum (E \Delta t) + K_f (P_{T+} - P_T) \quad (11.17)$$

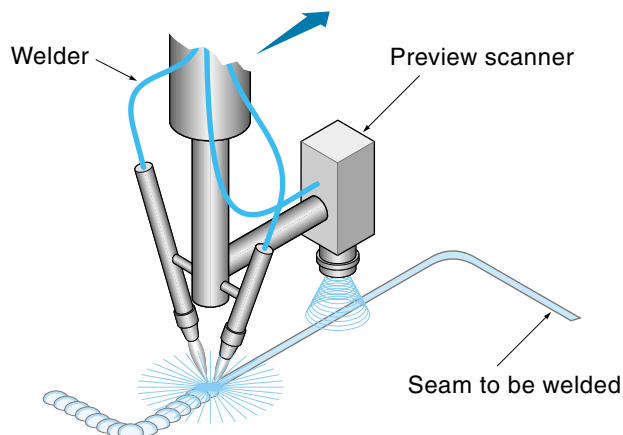
where

$K_p$  = proportional gain constant

$K_i$  = integral gain constant

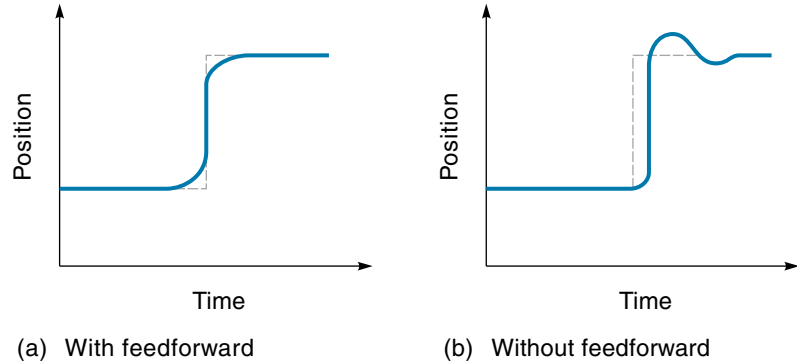
**Figure 11.35**

A welder using a preview sensor for PIP control.



**Figure 11.36**

Improved path control with feedforward.



$K_F$  = feedforward gain constant

$E$  = error ( $SP - PV$ )

$P_T$  = position it should be in now

$P_{T+1}$  = position it should be in, in the future (at  $T + 1$ )

Notice that the feedforward term,  $K_F (P_{T+1} - P_T)$ , is proportional to the difference between where the controlled object is and where it must be in the future. If this number is large, the system has a long way to go and should speed up. If the number is small or zero, the system will be stopping and so should begin slowing down. In each case, the value of the feedforward term is added to the controller output so that the controlled object is pushed with more or less force, depending on where it has to be in the near future. Figure 11.36 shows the path of two control systems, one with feedforward and one without. By anticipating the change in direction, the PIP system can begin slowing ahead of time and minimize overshoot.

## 11.8 FUZZY LOGIC CONTROLLERS

### Introduction

**Fuzzy logic**, a relatively new concept in control theory, is simply the acceptance of principles that have existed since the beginning of time: *Real-world quantities are not usually “all or nothing” or “black and white” but something in-between.* For example, if you are eating an apple bite-by-bite, at what point does it stop being an apple? If you are driving with the windows open at dusk, at what point does “refreshingly cool” turn into “a little chilly”? After the sun sets, at what point would you turn on the headlights? Traditional control systems can handle variations in input, but they handle it in a very one-dimensional way—that is, with a single mathematical model. For example, a proportional controller can respond to any error value between 0 and 100 %, but it handles all cases the same way, that is, by multiplying the error by  $K_p$ . Fuzzy logic controllers are modeled after the natural way *people* arrive at solutions:

- We apply different solution methodologies (rules), depending on the value of the stimulus. In other words, we might have two or three different types of response to the same general situation, but the specific response we choose will depend on the current stimulus.
- We frequently apply more than one of our “rules” at the same time to a single problem, so the actual course of action is the result of a combination of rules, each weighted differently according to the stimulus.
- We accept a certain amount of imprecision, which allows us to arrive at workable solutions to problems that are not completely defined and with much less processing time than it would take to arrive at an exact solution.

Let’s look at each of these in more detail. The first point: *We apply different solution methodologies to the same problem.* Another way to put this is that an individual has different ways of looking at the same general situation, and his or her response will be different depending on how it is “seen” at the time. For example, we can experience cooler temperatures as being in the “refreshing and brisk” category or “chilly” category, and we have a different range of possibly contradictory responses for each category—for example, in the first case, we may consider opening windows and breathing deeply; in the second case, we might consider putting on a sweater and turning up the heat.

The second point: *We often apply more than one of our rules to the same problem at the same time.* The course of action we end up taking is really a combination of two or more different response strategies, weighted according to the stimulus. For example, if the temperature is 68°F, you might process it as being *both* “refreshing” and “some-what cool,” and your response might be to breathe deeply while trying to remember where your sweater might be. When the temperature drops to 65°F, you might process it as being “too refreshing” (that is, less than 100% refreshing) and “definitely cool”; your response might be to find and put on the sweater.

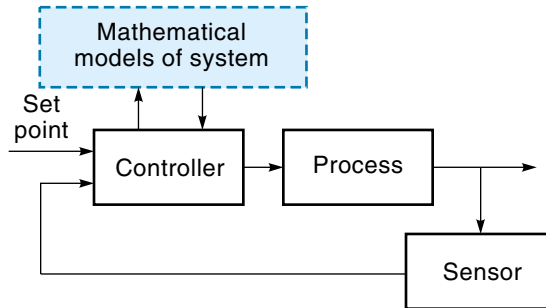
The third point: *We accept a certain amount of imprecision, which is very important in helping us arrive at workable solutions.* Consider the task of parking a car. Most people can park a car in a minute or two and do a “good enough” job—the car is somewhere between the cars to the front and back, and it is reasonably close to the curb. If the requirement was that you had to be exactly between the cars (to the nearest 0.1 in.) *and* exactly 12 in. from the curb, none of us could do it unless we installed precise sensors; even then, it might take months to maneuver the car to the *exact* place. Control systems (particularly servomechanisms) have traditionally been based on the most precise (and therefore complicated) mathematical model of the system one can make, which requires the controller to do much “number crunching.” However, the reality is that *an actual system is never an exact representation of its model* (real systems are riddled with nonlinearities that can never be completely accounted for); therefore, the controller’s time spent finding an exact solution to an inaccurate hypothetical model is time not well spent and does not necessarily lead to the best answer.

A fuzzy logic controller mimics the way a knowledgeable human operator would control something. That is, it applies a set of control rules appropriate to the situation,

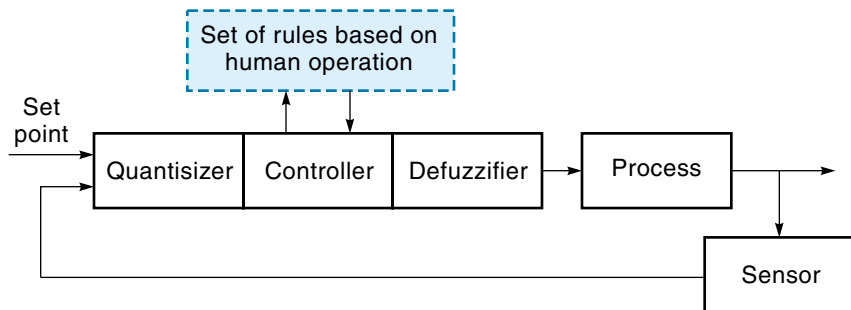


**Figure 11.37**

Traditional and fuzzy logic control block diagrams.



(a) Traditional feedback



(b) Fuzzy logic controller

which may overlap and even contradict each other. The final course of action is what we might call a *judgment*, which is some appropriate combination of all relevant factors. Figure 11.37(a) shows a block diagram of a traditional feedback control system. This controller makes its decisions about what to do based on either a mathematical model of the process or, in the case of a PID controller, a fixed set of mathematical relationships. Figure 11.37(b) shows a block diagram of a fuzzy logic control system. The fuzzy logic controller uses as its guide a set of response rules established by the knowledgeable operator or system engineer. The block marked “Quantisizer” takes the data from the sensor(s) and converts it into a form the fuzzy logic controller can use; for example, the data from a temperature sensor might be converted from degrees into **fuzzy predicates** such as *brisk*, *cool*, *cold*, and *very cold*.<sup>\*</sup> The output of the fuzzy logic controller is actually a set or range of responses. (Think back at how often you have had “mixed feelings” about a response to a problem.) However, it is clear that an actuator needs a specific signal, such as a voltage for a motor, so it is the job of the “defuzzifier” block [Figure 11.37(b)] to distill the controller’s multilevel response into a specific, “crisp,” actuator control signal.

<sup>\*</sup>Other possibilities are *fuzzy truth values* (“true,” “quite true,” “not very true”), *fuzzy quantifiers* (“most,” “few,” “almost”), and *fuzzy probabilities* (“likely,” “very likely,” “not very likely”).

Fuzzy logic was first proposed by L. A. Zadeh working at Berkeley in 1965. However, Japanese industry really embraced the idea and developed applications in the area of fuzzy logic control. One of the first major applications of fuzzy logic control was the subway system in the city of Sendai. Performance of this system turned out to be much improved over traditional controllers. For example, the cars stop with the doors almost exactly where they should be, and the acceleration is so smooth that patrons hardly have to use the overhead handrail. Other examples are the shifting capabilities in automatic transmissions (Nissan, Honda, and Saturn to name a few). The Nissan system claims to cut fuel consumption by 12-17%. A traditional automatic transmission is built to always shift at certain speeds and engine conditions, which can lead to a lot of shifting. Human drivers not only don't shift as often but also don't always shift at the same speed, for instance, when climbing a hill. Fuzzy logic-controlled washing machines adjust the amount of water, amount of detergent, and cycle time to how dirty and how many clothes are in the load.

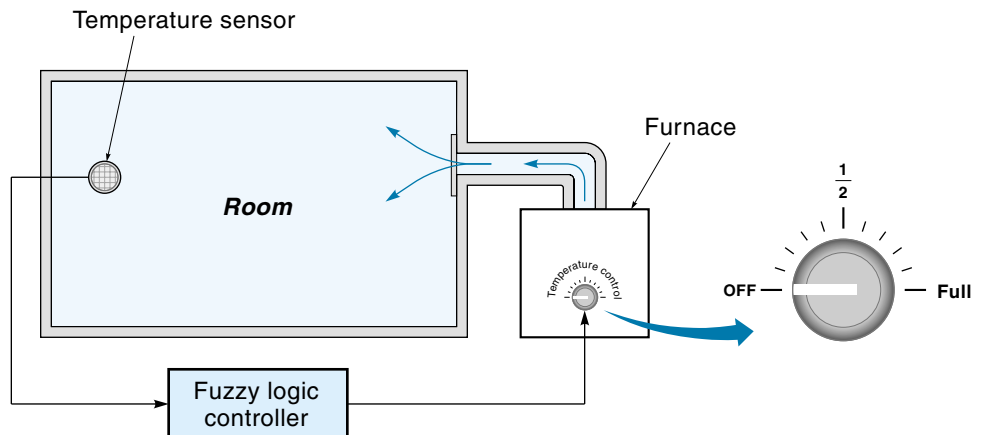
### Example of a One-Input System

To understand how the fuzzy logic controller actually works, let's look at a simple example of controlling the temperature in a room. The system consists of a room, a gas furnace, a temperature sensor, and the fuzzy logic controller (Figure 11.38). The gas flow to the furnace is regulated by a knob that turns from off to full on (in ten notches). The fuzzy controller will sample the temperature and, then based on a set of rules, adjust the gas knob up or down. The controller is trying to maintain a comfortable temperature (defined as 70°F) in the room.

For simplicity, we will define only three temperature conditions: *warm*, *medium*, and *cool*. Each condition, represented on the graph of Figure 11.39(a), looks like a triangle and is called a **membership function** because it represents a range of temperatures that are all members of the same category. For example, we're willing to say that

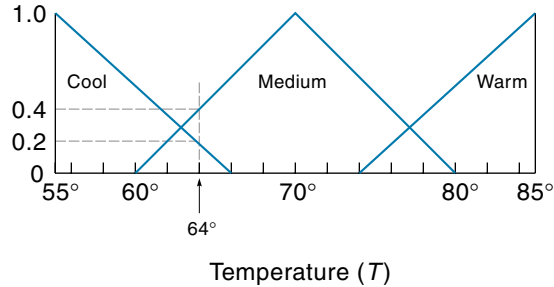
**Figure 11.38**

Using a fuzzy logic controller to control the temperature in a room.

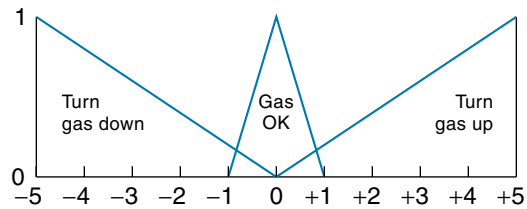


**Figure 11.39**

Input and output sets for a one-input fuzzy logic controller.



(a) Input sets for three temperature conditions



Number of notches on gas knob to turn up or down

(b) Output sets for three kinds of action

(for humans) *medium* can be anything between 60 and 80°, but it is most true at 70°. At the same time, we are saying that any temperature below 66° is *cool*; however, using the term *cool* is *more true at 55° than it is at 66°*. The terms *medium* and *cool* are called **fuzzy sets** because they express a range of values. Notice that the fuzzy sets overlap (for example, between 60° and 66°). This simply means that a specific temperature can be a member of more than one set; for example, 64° is a “little bit cool” and “some-what medium” at the same time.

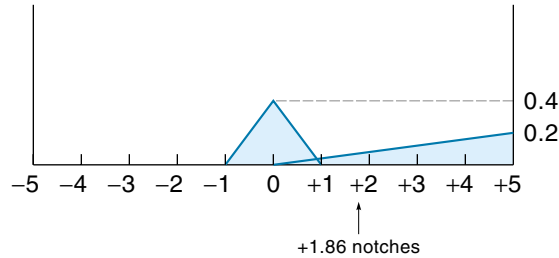
The fuzzy controller operates on a set of if-then rules, which are stated in very linguistic terms. In our example, we might have three rules:

- Rule 1: If the temperature is *cool*, then turn up the gas.
- Rule 2: If the temperature is *medium*, then the gas is OK.
- Rule 3: If the temperature is *warm*, then turn down the gas.

Next we have to define “turning up” and “turning down” the gas. Figure 11.39(b) shows this in another graph of fuzzy sets. Here you can see that the triangle for “turn gas down” corresponds to turning the gas knob toward off, anywhere from one to five notches. “Gas OK” corresponds to not moving the knob, or at most moving it one notch in either direction, and so on. With these two graphs and the set of rules, the fuzzy logic controller can operate. For example, if the temperature sensor reported 64°, the quantizer would determine that 64° is 20% *cool* and 40% *medium* [see Figure 11.39(a)].

Figure 11.39

Continued.



(c) Output weighted by input conditions, showing defuzzified output to be 1.86 notches

This means that Rule 1 applies at 20%, Rule 2 applies at 40%, and Rule 3 does not apply at all. We now redraw the output graph with the magnitudes adjusted per the rules [Figure 11.39(c)]. That is, Rule 1 says, “If the temperature is *cool*, then turn up the gas”; thus, because Rule 1 is 20% true, we set the maximum value of “turn up the gas” at 20%. Similarly, Rule 2 is 40% true, so we adjust the “Gas OK” triangle to a 40% maximum value. Technically, the output of the fuzzy logic controller is the union of these two sets [Figure 11.39(c)]. However, the gas knob needs a specific command, so we must “defuzzify” the output set, that is, find some average value. In this case, we find the point on the horizontal axis where the area under the curve is the same on either side. This point is 1.86, meaning we should turn up the gas knob 1.86 notch. Does this make sense? Yes—a temperature of  $64^\circ$  is a little cool, so you would expect the controller to turn up the gas a little.

### Example of a Two-Input System

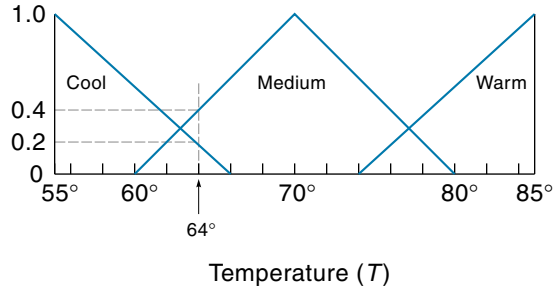
Let’s look at a more sophisticated fuzzy logic controller that uses two inputs. Using again the example of a room-temperature controller, let’s add a second input: “rate-of-temperature change.” So now, the controller knows what the temperature ( $T$ ) is and how fast it is changing ( $\Delta T$ ). As before, the temperature input will be evaluated as being either *warm*, *medium*, or *cool* [Figure 11.40(a)]. We also define  $\Delta T$  in three simple categories: *lowering*, *steady*, and *raising* [Figure 11.40(b)]. For example, *lowering* is defined as the temperature falling at a rate of  $0.2^\circ$  per minute to  $1^\circ$  per minute. On the graph in Figure 11.40(b), the categories are represented as triangles, where the height represents the extent to which that term is true, going from a probability of 0 to a probability of 1 (or from 0 to 100%).

Now we address the important step of establishing the rules for the controller. There are now nine possible sets of inputs  $T$  and  $\Delta T$ , as given below. For each set of fuzzy inputs, a fuzzy output is specified. Notice that these rules are just simple common sense.

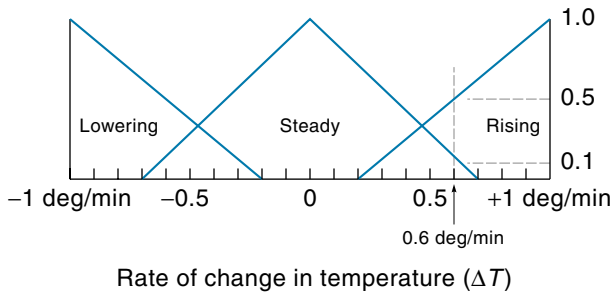
- Rule 1: If  $T$  is *cool* and *lowering*, then increase the gas sharply.
- Rule 2: If  $T$  is *cool* and *steady*, then increase the gas.
- Rule 3: If  $T$  is *cool* and *raising*, then the gas is OK.

**Figure 11.40**

Input sets and an output table for a two-input fuzzy logic controller.



(a) Input sets for three temperature conditions



(b) Input sets for three kinds of action

		$\Delta T$		
		Lower	Steady	Raise
$T$	Cool	+5 Rule 1	+2 Rule 2	0 Rule 3
	Medium	+2 Rule 4	0 Rule 5	-2 Rule 6
	Warm	0 Rule 7	-2 Rule 8	-5 Rule 9

(c) Output table

- Rule 4: If  $T$  is *medium* and *lowering*, then increase the gas.
- Rule 5: If  $T$  is *medium* and *steady*, then the gas is OK.
- Rule 6: If  $T$  is *medium* and *raising*, then decrease the gas.
- Rule 7: If  $T$  is *warm* and *lowering*, then the gas is OK.
- Rule 8: If  $T$  is *warm* and *steady*, then decrease the gas.
- Rule 9: If  $T$  is *warm* and *raising*, then decrease the gas sharply.

Next, we must assign values to the output conditions of “increase gas,” “gas is OK,” and “decrease gas.” Each output condition could be a fuzzy set (as they were in the previous example), but this time we will use a table of discrete values [Figure 11.40(c)]. Each position in the table corresponds to a rule and contains a number from 0 to  $\pm 5$ , which corresponds to the number of notches to turn the gas knob up or down. For example, if the temperature is *cool* and *lowering* (Rule 1), then turn the gas way up (five notches). If the temperature is *cool* and *steady*, then turn the gas up a little (two notches), and so on. With the system thus defined, we can now use an example to show how it works.

Assume that the temperature sensor reports  $64^\circ$  (same as previous example) and that the temperature is rising somewhat rapidly at  $0.6^\circ$  per minute. From the graph of temperature-input sets [Figure 11.40(a)], you can see that  $64^\circ$  corresponds to 20% *cool* and 40% *medium*. Then, from the  $\Delta T$  graph [Figure 11.40(b)] you can see that  $0.6^\circ$  per

minute corresponds to 10% *steady* and 50% *raising*. Looking at the list of nine rules, we see that only four rules apply in this case:

- Rule 2: If  $T$  is *cool* and *steady*, then increase the gas.
- Rule 3: If  $T$  is *cool* and *raising*, then the gas is OK.
- Rule 5: If  $T$  is *medium* and *steady*, then the gas is OK.
- Rule 6: If  $T$  is *medium* and *raising*, then decrease the gas.

Because the variables are fuzzy, each rule will contribute to a different degree. The first step toward arriving at a single output value is to compute the compatibility for each rule. The *compatibility* is simply the product of the two probabilities in the rule, as expressed in Equation 11.18:

$$\text{Compatibility} = w_i = A_{i1} \times A_{i2}, \quad i = 1, 2, \dots \quad (11.18)$$

where

$w_i$  = compatibility (a measure of the influence of Rule  $i$ )

$A_{i1}$  = probability of first condition ( $T$ )

$A_{i2}$  = probability of second condition ( $\Delta T$ )

Applying Equation 11.18 to the applicable rules of this example,

$$w_2 = 0.2 \times 0.1 = 0.02^*$$

$$w_3 = 0.2 \times 0.5 = 0.1$$

$$w_5 = 0.4 \times 0.1 = 0.04$$

$$w_6 = 0.4 \times 0.5 = 0.2$$

Now, using Figure 11.40(c), we can determine the output value for each applicable rule:

$$y_i = w_i \times B_i, \quad i = 1, 2, 3, \dots \quad (11.19)$$

where

$y_i$  = output from a single rule

$w_i$  = compatibility (a measure of the influence of Rule  $i$ )

$B_i$  = appropriate value from the output table

Applying Equation 11.19 to Rules 2, 3, 5, and 6,

$$y_2 = w_2 \times B_2 = 0.02 \times 2 = 0.04$$

$$y_3 = w_3 \times B_3 = 0.1 \times 0 = 0$$

$$y_5 = w_5 \times B_5 = 0.04 \times 0 = 0$$

$$y_6 = w_6 \times B_6 = 0.2 \times -2 = -0.4$$

\*That is, for Rule 2,  $cool = 0.2$  and  $steady = 0.1$ .

We now have four outputs from the four rules, but we need to defuzzify this data into a single command for the gas knob. A simple average would not work because some conditions are evoked more strongly than others. One approach is to use a *weighted mean*, as specified in Equation 11.20:

$$y_{\text{tot}} = \frac{w_1 y_1 + w_2 y_2 + \dots}{w_1 + w_2 + \dots} \quad (11.20)$$

Applying Equation 11.20 to this problem,

$$\begin{aligned} y_{\text{tot}} &= \frac{w_2 y_2 + w_3 y_3 + w_5 y_5 + w_6 y_6}{w_2 + w_3 + w_5 + w_6} \\ &= \frac{(0.02 \times 0.04) + (0.1 \times 0) + (0.04 \times 0) + (0.2 \times -0.4)}{0.02 + 0.1 + 0.04 + 0.2} \\ &= \frac{-0.079}{0.36} = -0.22 \text{ notch} \end{aligned}$$

Therefore, the final output of the fuzzy logic controller specifies that the gas should be turned down slightly (0.22 notch). However, the temperature in the room is actually somewhat cool (64°); why does the controller send a signal to turn down the gas? It is mimicking our intelligence, and *we* know that with the temperature rising so fast (0.6° per minute), we had better ease the heat back now or it will surely overshoot.

## Closing Thoughts

It seems certain that fuzzy logic controllers will be used in more and more products and systems. On the consumer side, they are already being used in automobiles, camera focusing, air conditioners, and microwave ovens, to name a few. For industrial systems, they are being used in process control for temperature and flow control, and in servo-mechanisms for speed control and helicopter autopilots. Also, fuzzy logic controllers are being used in conjunction with traditional PID controllers, where the job of the fuzzy controller is to adapt the PID parameters to changing conditions.

In many cases, the fuzzy controller is a traditional microcontroller, such as the Intel 8051 or Motorola 68HC11, which has been programmed to implement fuzzy logic. Also, new fuzzy logic control ICs that are designed specifically for this application are becoming available.

A final word: Although fuzzy logic controllers have repeatedly performed better than their traditional control system counterparts, these improvements do not come without effort. Finding the right rule set and specifying the nature and range of the fuzzy variables can be very time-consuming. Arriving at the right set of inputs for the subway train in Sendai took engineers months of tuning. However, like the traditional digital controller, the fuzzy logic control algorithm is implemented in software, not hardware, so fine-tuning can be done with minimum system downtime.

## SUMMARY

The controller, a part of the control system, directs the actuator to move some parameter, which is called the controlled variable. A feedback controller has two inputs, the set point ( $SP$ ) and the process variable ( $PV$ ), and one output. The set point specifies the desired position of the controlled variable. The process variable is the actual position of the controlled variable, as reported by a sensor. The error value ( $E$ ) is the difference between the set point and the process variable ( $E = SP - PV$ ). Based on the error, the controller creates a control signal for the actuator.

There are two major classifications of feedback control systems: servomechanisms and process control. Servomechanisms are usually mechanical motion systems. Analysis of some servomechanisms may require a highly mathematical approach (not presented in this text). Process control refers to a system that attempts to keep some process at a constant value. Process control systems are usually slower than servomechanisms, and empirical data may need to be used in the design.

There are different levels of control strategy. The simplest are the on-off controllers; with these, the actuator is either on full force or off, as in a thermostatically controlled heating system. Proportional control is more sophisticated; with this system, the controller provides a restoring force that is proportional to the error. One problem with proportional control is that it cannot reduce the error to zero because it provides no force at zero error.

Integral control is sometimes added to proportional control and provides a restoring force based on the sum of past errors. Integral control can reduce the steady-state error to zero, but it may introduce stability problems.

Derivative control, which can also be added to proportional control, provides a restoring force that is based on the rate that the error is increasing (or decreasing) and tends to quicken the response and reduce overshoot.

PID control combines proportional, integral, and derivative controls into one controller and is a very common control strategy, especially for process control systems. The PID controller is tuned to perform in any particular application by adjusting the gain constants for proportional, integral, and derivative. Analog PID controllers use differential amplifiers to create and then sum the three control components. Digital PID controllers work under program control; the program is in the form of an endless loop. With each pass through the loop, the following functions are performed: Read in the process variable and set point, calculate the controller output, and send output to the actuator.

Fuzzy logic controllers are relatively new and use a completely different approach than traditional controllers. Fuzzy logic controllers are not based on a mathematical model of the system but instead implement the same control “rules” that a skilled human operator would. Fundamental in the concept of fuzzy logic is the recognition that rules and conditions come in degrees, as specified in linguistic terms—for example, *warm*, *warmer*, and *very warm*.



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## GLOSSARY

**adaptive controller** A controller, such as a PID, with the additional ability to monitor and improve its own performance.

**aliasing** The event that occurs when the wrong data are reconstructed from the sampled sensor data because the sample rate is too slow.

**autotuning** The ability of some digital controllers to continuously fine-tune themselves by making small changes in their gain constants, based on past performance.

**bias** A value that is added to the output of the controller to compensate for a constant disturbing force, such as gravity.

**Bode plot** A graph that plots system gain and phase shift versus frequency; it can be used to determine system stability.

**closed-loop system** A control system that uses feedback—that is, a sensor tells the controller the actual state of the controlled variable.

**continuous-cycle method** A method of tuning a PID controller that involves driving the controller into oscillations and measuring the response; the PID constants are computed based on these data.

**controlled variable** The parameter that is being controlled, such as temperature in a heating system.

**dead band** The small range, on either side of the set point (of the controlled variable), where the error cannot be corrected by the controller because of friction, mechanical backlash, and the like.

**dead time** The interval of time between the correcting signal sent by the controller and the response of the system.

**dead zone** *See* **dead band**.

**derivative control** A feedback control strategy where the restoring force is proportional to the rate that the error is changing (increasing or decreasing); derivative control tends to quicken the response and reduce overshoot.

**error ( $E$ )** The difference between where the controlled variable should be and where it actually is ( $E = SP - PV$ ).

**feedforward** The concept of letting the controller know in advance that a change is coming.

**fuzzy logic** A new control strategy, modeled originally on human thought processes, where decisions are made based on a number of factors and each factor can have such qualities as *maybe* and *almost*, to name only two.

**fuzzy predicates** Linguistic terms of degree, used in describing actual variables—for example, *warm*, *very warm*, *not so warm*, and so on.

**fuzzy set** The range of one fuzzy predicate in a system—that is, all values of temperature that correspond to *very warm*.

**gain** A multiplication factor; a steady-state transfer function. The term *system gain* refers to the proportional gain factor ( $K_p$ ).

**gain margin** A form of a safety margin. When the phase lag is  $180^\circ$ , the gain margin is the amount the gain could be increased before the system becomes unstable.

**integral control** A feedback control strategy where the restoring force is proportional to the sum of all the past errors multiplied by time. This type of control is capable of eliminating steady-state error but increases overshoot.

**iteration** *See scan.*

**lag time** The time between the change of the set point and the actual movement of the controlled variable.

**membership function** Used in conjunction with fuzzy logic control, the distribution of values that have been grouped into one category.

**on-off control** *See two-point control.*

**open-loop system** A type of control system where there is no feedback; consequently, the controller does not know for sure the exact condition of the controlled variable.

**overshoot** The event that occurs when the path of the controlled variable reaches and then goes past the set point, before turning around or stopping.

**phase margin** A form of a safety margin. When the system gain is 1, the phase margin is the amount the phase lag could be increased before the system becomes unstable.

**PID** Stands for Proportional + Integral + Derivative; a control strategy that uses proportional, integral, and derivative feedback.

**PIP** Stands for Proportional + Integral + Preview; a control strategy that uses feedforward so that it knows in advance if a change is coming.

**positive feedback** The event that occurs when the feedback signal from the sensor is added to the set point (instead of subtracted); positive feedback will likely cause the system to oscillate.

**process control system** The type of system where the controller is attempting to maintain the output of some process at a constant value.

**process variable (PV)** The actual value of a variable, as reported by the sensor.

**proportional band** A term used to describe the gain of a proportional control system. It is the change in error that causes the controller output to change its full swing.

**proportional control** A feedback control strategy where the controller provides a restoring force that is proportional to the amount of error.

**PV** *See* process variable.

**reaction-curve method** A method of tuning a PID controller that involves opening the feedback loop, manually injecting a step function, and measuring the system response. The PID constants are computed based on these data.

**rise time** The time it takes for the controlled variable to rise from 10 to 90% (of its final value).

**sample rate** The times per second (or per minute) that a sensor is read; for a digital controller, the sample rate usually corresponds to the scan rate.

**scan** One pass through the program of a digital-type controller.

**servomechanism** A mechanical-motion system that uses feedback control, such as a robot.

**set point (SP)** An input into the controller, it represents the desired value of the controlled variable.

**settling time** The time it takes for the system response to settle down to some steady-state value.

**SP** *See* set point.

**steady-state error** The error that remains after the transient response has died away.

**step change** The event that occurs when a discrete change is made to the set point.

**three-position control** Similar to the two-point system except the actuator can exert force in two directions, so the system is either off-up-down, off-right-left, and so on.

**transient response** The actual path that the controlled variable takes when it goes from one position to another.

**two-point control** A simple type of control system where the actuator is either on full force or off; also called *on-off control*.

**tuning** When applied to a control system, this term refers to the process of making adjustments to the gain constants  $K_p$   $K_I$   $K_D$  until system performance is satisfactory.

**windup** Associated with integral feedback, the problem that occurs when too much error accumulates as a result of the system saturating before it can achieve the state that it theoretically should.

## EXERCISES

### Section 11.1

1. Assume that Figure 11.25(a) is the response curve of a control system to a step function. Determine the following parameters: rise time, overshoot, and settling time.
2. A robot arm was commanded to go to a new position. Its response was recorded and is shown in Figure 11.41. Determine the rise time, overshoot, settling time, and steady-state error of the response.

### Section 11.2

3. Give an example of a two-point control system (other than a heating system) and explain how it works.
4. A thermostat is used to maintain a temperature of  $87^\circ$  F. An accurate recording of the room temperature is shown in Figure 11.42. The respective cut-in and cut-off points are currently  $84^\circ$  and  $90^\circ$ . What would you predict the cycle time would be if the cut-in and cut-off points were respectively changed to  $86^\circ$  and  $88^\circ$ ?
5. Describe how three-position control might be used to cause a windmill to always face into the wind. The base of the windmill is rotated with a bidirectional DC motor.

### Section 11.3

6. In your own words, explain the principle of proportional control.
7. The system gain for the position system shown in Figure 11.6 is  $2 \text{ ft} \cdot \text{lb}/\text{deg}$ . If the set point is  $50^\circ$ , find the torque on the arm when the arm is at  $15^\circ$  and  $45^\circ$ .
8. A proportional controller for a rotating antenna has a gain  $K_p$  of  $5 \text{ in.} \cdot \text{oz}/\text{deg}$ . The antenna was initially pointing due south but was then commanded to point southeast.
  - a. Find the initial torque supplied to the antenna.
  - b. What is the torque when the antenna gets to within  $5^\circ$  of its set point?

Figure 11.41

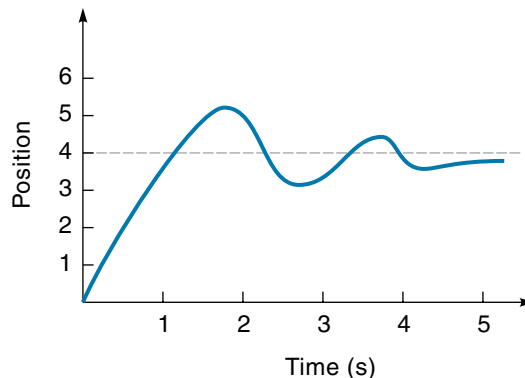
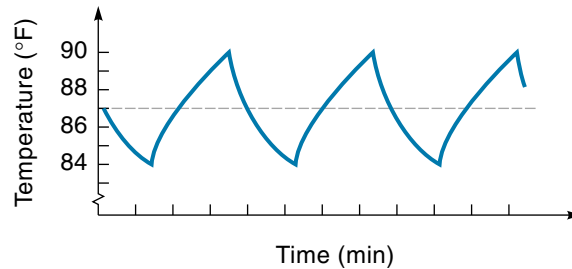


Figure 11.42



9. Explain why a proportional control system lifting a weight can never reduce the steady-state error to zero.
10. A proportional control system with a  $K_p$  of 5 ft · lb/deg controls a robot arm that is 2 ft long. What steady-state error would you expect if the robot lifted a 10-lb object?
11. A linear-motion proportional control system has a gain of  $K_p = 5$  lb/in. A friction force acts on the controlled variable with a constant force of 0.25 lb. Find the length of the dead band.
12. A proportional control system is to be used to control the wing flaps of a jet airplane. The wind load is expected to cause as much as 600 in. · lb of torque. The system should be able to keep the flaps to within  $5^\circ$  of the set point. Find the gain  $K_p$  necessary to meet the requirements.

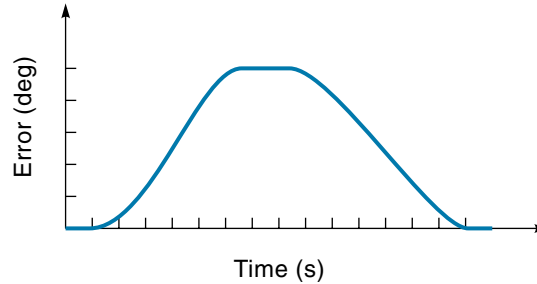
### Section 11.4

13. Explain how the addition of integral feedback in a proportional control system eliminates steady-state error.
14. A control system with integral control has a  $2^\circ$  steady-state error similar to the graphs in Figure 11.15. The steady-state error is purely the result of a 7 in. · oz friction force. How many seconds will it take the integral control to correct this error? (Assume  $K_I K_p = 1$  in · oz/deg · s)
15. A proportional control system is working against 20 in. · lb of friction that causes a steady-state error of  $5^\circ$ . If integral feedback is added with a  $K_I$  of 2/s, how long would it take for the error to be eliminated?
16. What problem is solved and what new problem is created with the addition of integral feedback?

### Section 11.5

17. Derivative feedback produces a force that is proportional to what?
18. A plot of error versus time (for a control system) is shown in Figure 11.43.  $K_p = 1$  in · oz/deg and  $K_D = 2$  s Find the maximum positive and negative values of derivative output. Make a sketch of the derivative output versus time.
19. Explain how derivative feedback makes a control system more responsive to rapid change and how it reduces overshoot.

Figure 11.43



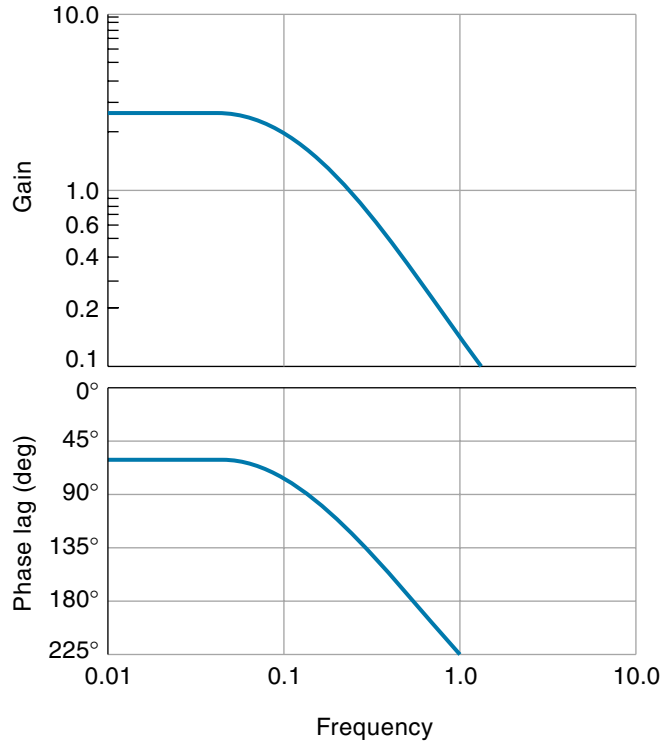
### Section 11.6

20. Explain how the three elements of the PID control system work together to create a practical control system.
21. Draw a block diagram for an analog PID controller, indicating the function that each block performs.
22. How does a digital controller function? List the general steps that a digital controller program follows.
23. How does a digital controller integrate the error signal curve?
24. How does the digital controller find the derivative of the error signal?
25. Write a program in BASIC (or any other language) to implement a PID controller. The sample time is 0.5 s. Assume that the set point and error data are constantly available from READ or INPUT statements. Assume  $K_p = 2$ ,  $K_I = 1.5$ ,  $K_D = 1.8$ , and all units are compatible. Send out the controller drive signal with an OUT or a PRINT statement.
26. The Bode plot for a control system is shown in Figure 11.44. Is this system stable? Find the gain margin and phase margin.
27. A system oscillates with a period of 5 s. There is a 1-s lag time between the controller output and the controlled-variable movement. How many degrees of phase lag does this system have under these conditions?
28. A control system is to be tuned using the continuous-cycle method. It was found that the system went into oscillation when  $K'_p$  was 0.3 V/deg. Figure 11.45 shows the system response. Find  $K_p$ ,  $K_I$  and  $K_D$ .
29. A control system is to be tuned using the reaction-curve method. The test input and resulting output are shown in Figure 11.46. Find  $K_p$ ,  $K_I$  and  $K_D$ .
30. A control system samples a position 30 times per second. If the position signal is cycling back-and-forth, what is the highest frequency the controller can follow? Give both theoretical and practical values.
31. Is a high sample rate good or bad for a control system? Explain.

### Section 11.7

32. What is the necessary condition that allows PIP control to be used?
33. What is *feedforward*, and why does using this concept improve path control over the PID system?

Figure 11.44



**Section 11.8**

- 34. For the single-input fuzzy logic temperature controller specified in Figures 11.38 and 11.39, determine the defuzzified output for a temperature of 76°. Does your answer make sense?
- 35. For the single-input fuzzy logic temperature controller specified in Figures 11.38 and 11.39, determine the defuzzified output for a temperature of 62°. Does your answer make sense?

Figure 11.45

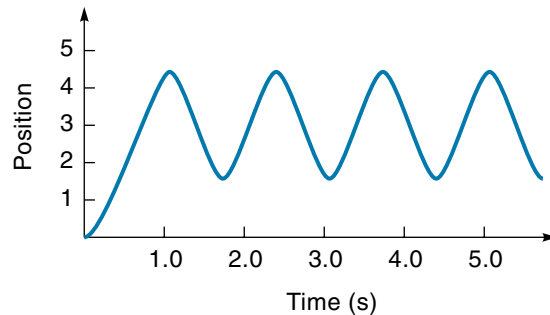
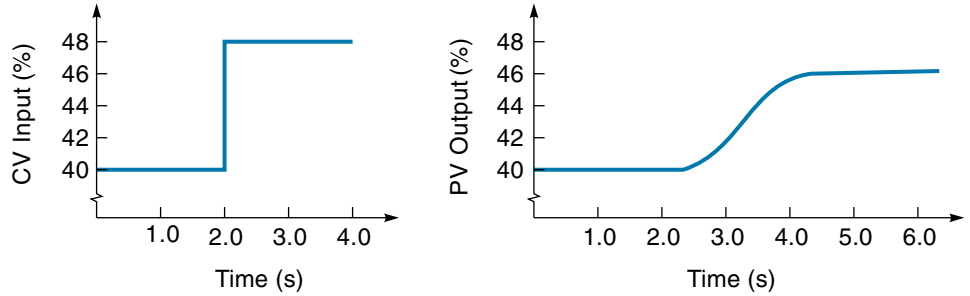


Figure 11.46



36. For the two-input fuzzy logic temperature controller specified in Figure 11.40, determine the defuzzified output for a temperature of  $76^{\circ}$  and  $\Delta T$  of  $-0.3$  deg/ min. Does your answer make sense?
37. For the two-input fuzzy logic temperature controller specified in Figure 11.40, determine the defuzzified output for a temperature of  $62^{\circ}$  and  $\Delta T$  of  $+0.7$  deg/ min. Does your answer make sense?



# CHAPTER 12

## Relay Logic, Programmable Logic Controllers, and Motion Controllers

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### OBJECTIVES

After studying this chapter, you should be able to:

- Explain the operation of electromechanical relays, time-delay relays, counters, and sequencers.
- Explain the purpose and operation of a ladder diagram.
- Explain the operation of a relay-based controller.
- Understand the concept and purpose of a programmable logic controller (PLC).
- Understand the hardware and wiring required in a PLC-based system.
- List the steps that must be taken to make a PLC control system operational.
- Understand the basic instructions used in a PLC program.
- Differentiate the ways that a PLC can be programmed.
- Understand how PLCs are used with networks.
- Explain the purpose and operation of a motion controller.

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### INTRODUCTION

The automatic control of repetitious mechanical or physical processes is a common control problem, which abounds in major appliances, office machines, the manufacturing industry, and industry in general. Traditionally, this kind of process control was done with electromechanical devices such as relays, timers, and sequencers. With this approach, however, the circuit must be rewired if the control logic changes, which was a particular problem to the automotive industry because of the annual model changeovers. In response to this problem, in the late 1960s, General Motors developed the specifications for a programmable electronic controller that could replace the hard-wired relay circuits. Based on those specifications, Gould Modicon Company developed the first programmable logic controller (PLC). The PLC is a small, microprocessor-based process-control computer

that can be connected directly to such devices as switches, small motors, relays, and solenoids, and it is built to withstand the industrial environment. This chapter will introduce the principles of relay logic control and then describe the general operation and programming of the PLC. For those who want a detailed working knowledge of a PLC, a tutorial (published by Allen-Bradley) for connecting, programming, and operating the Allen-Bradley MicroLogix 1000 and SLC 500 PLCs, is included in the appendix. The chapter concludes with discussions on PLC networks and motion controllers.

## 12.1 RELAY LOGIC CONTROL

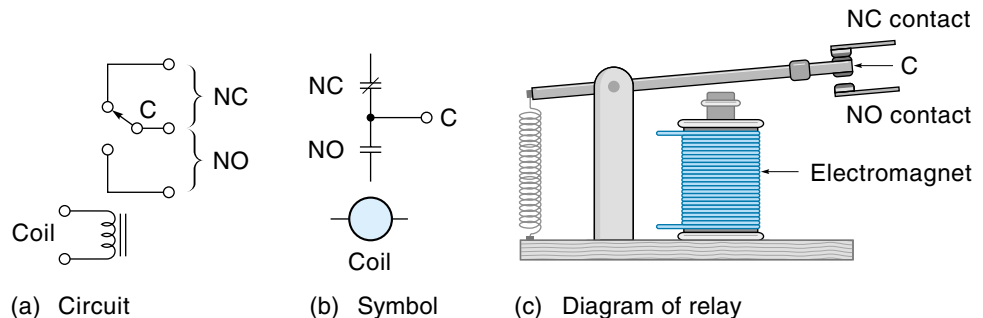
### Relay Logic

Relays as a device are covered in Chapter 4; however, a quick review of the basics is appropriate here. An **electromechanical relay** (EMR) is a device that uses electromagnetic force to close (or open) switch contacts—in other words, an electrically powered switch. Figure 12.1 shows a diagram of a relay. Relay contacts come in two basic configurations—**normally open contacts** (NO), which are open in the unenergized state (and close when the relay coil is energized), and **normally closed contacts** (NC), which are closed in the unenergized state (and open when the relay is energized). By convention, *the relay symbol always shows the contacts in the unenergized state*. Relays are available with a variety of multiple-contact configurations, two of which are shown in Figure 12.2.

No doubt, relays were first developed to satisfy two electrical control needs: remote control (the ability to turn devices on and off from a remote location) and power amplification. An example of power amplification is the starting relay in a car (a low-current ignition switch energizes a relay, which in turn passes a high current to the starting motor). Electrical designers soon recognized that relays could be used to implement control logic. AND, OR, and NOT gates, as well as flip-flops, can be wired from relays (Figure 12.3). For the AND gate in Figure 12.3(a), relays A *AND* B must be energized for a voltage to be at X. Figure 12.3(b) shows an OR gate. Here, if either relay A *OR* B is energized, the output X receives a voltage. Figure 12.3(c) shows a relay NOT gate. If relay A is energized, the output is 0 V; if it is *NOT* energized, the output receives a voltage.

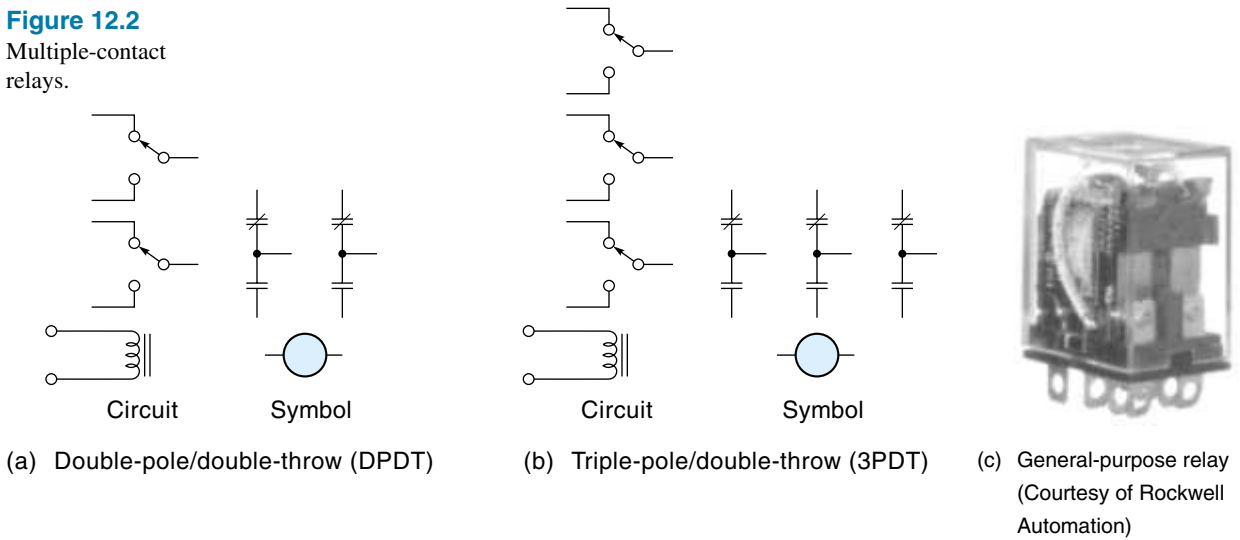
**Figure 12.1**

A relay: showing normally closed (NC) and normally open (NO) contacts.



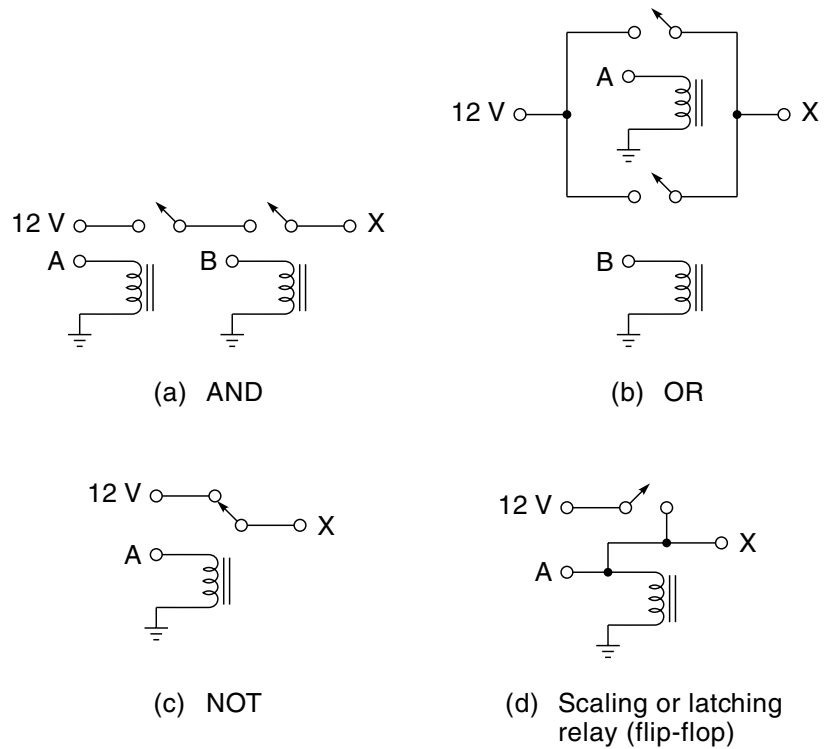
**Figure 12.2**

Multiple-contact relays.



**Figure 12.3**

Logic functions from relays.



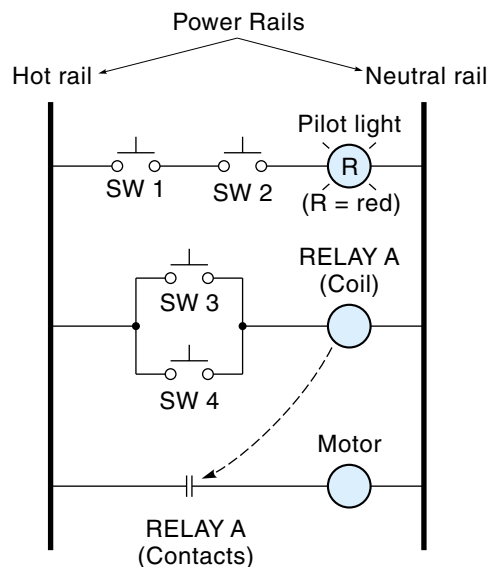
A flip-flop can be made from a relay using a principle called **sealing**. As seen in Figure 12.3(d), once the relay is energized, an electrical path through the contacts takes over the job of providing power to the coil, and the original A voltage can be removed. (This process is frequently called **latching**. However, a relay catalog uses the term *latching relay* to mean a relay that has a built-in mechanical-latching mechanism.) To unseal the relay, the power coming through the contacts must be (temporarily) removed.

## Ladder Diagrams

Switches and relays became widely used in industry for controlling motors, machines, and processes. A switch can turn a single machine on and off, but a relay logic network can control an entire process—turning on one machine, waiting until that operation is done, then turning on the next operation. Decisions can be made by the logic—for example, if a part is too tall, it goes in one bin; otherwise, it goes in the other bin.

Eventually, the **ladder diagram**, a special type of wiring diagram, was developed for relay-and-switching control circuits. Figure 12.4 shows a ladder diagram (notice it resembles a ladder). The ladder diagram consists of two **power rails**, which are placed vertically on each side of the diagram, and **rungs**, which are placed horizontally between the power rails. The power rails are the source of power in the circuit (AC or DC), where the left rail is the “hot” side (voltage) and the right rail is neutral (AC) or ground (DC). Therefore, each rung is connected across the voltage source and is an independent circuit. A rung typically contains at least one set of switch or relay contacts and usually only one load such as a relay coil or motor. When the contacts in a particular rung

**Figure 12.4**  
A relay logic  
ladder diagram.



close to make a continuous path, then that rung becomes active, and its load is energized. For example, the top rung in Figure 12.4 contains two switches and a pilot light in series. For the rung to become active, both switches must close, which will then apply voltage to the light. The middle rung has two switches in parallel, so only one of these switches must be closed to make the rung active. The load in this rung is a relay coil (RELAY A). The bottom rung contains a set of NO contacts from RELAY A, and the load is a motor. Consequently, when either SW 3 or SW 4 of the middle rung closes, RELAY A coil is energized, and the motor (in the bottom rung) starts.

### EXAMPLE 12.1

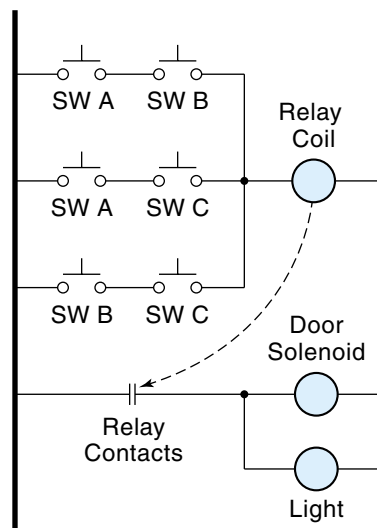
In a certain bank, each of three bank officers has a unique key to the vault. The bank rules require that two out of the three officers be present when the vault is opened. Draw the ladder diagram for a relay logic circuit that will unlatch the door and turn on the light when two of the three keys are inserted.

### SOLUTION

Three combinations of keys will open the vault: A and B, A and C, and B and C. Each of the three keys—A, B, and C—fits in its own key switch that has two sets of NO contacts. Figure 12.5 shows the completed ladder diagram. The top rung has three branches of switch contacts, one for each acceptable possibility. At least one branch must have continuity so that the relay coil is energized. The bottom rung, activated when the relay contacts close, provides power to the door-latch solenoid and the vault light.

**Figure 12.5**

A ladder diagram  
(Example 12.1).



**EXAMPLE 12.2**

A simple pick-and-place robot picks up parts from one conveyer belt and places them on another belt, as shown in Figure 12.6(a). When a part moving along the lower conveyer belt activates Switch 1, a solenoid-powered gripper clamps on the part and carries it toward the upper conveyer belt. When the gripper reaches Switch 2, it releases the part and moves back (empty) to receive the next part. When the gripper reaches Switch 3, it halts and waits for the next part to start the cycle all over again. Draw the relay logic ladder diagram to control this operation.

**SOLUTION**

Figure 12.6(b) shows the completed ladder diagram. The cycle starts when a part on the lower conveyer belt activates Switch 1—momentarily closing contacts *SW 1-1* (shown in the top rung). This action energizes the *MOTOR* relay, and the motor starts receiving current through the *M-1* contacts (bottom rung). The *MOTOR* relay is sealed (latched) via relay contacts *M-2* (now even when *SW 1-1* opens, the *MOTOR* relay will stay energized through the *M-2* contacts).

While the motor is getting started, the second set of Switch 1 contacts (*SW 1-2*) in rung 2 energizes the *DIRECTION* relay, causing the motor to drive the gripper toward the upper conveyer belt (the *DIRECTION* relay determines the motor direction with DPDT contacts *D-3*). Using the same technique as in rung 1, the *DIRECTION* relay is sealed “on” through one of its own contacts (*D-1*).

When the gripper arrives at the upper conveyer belt, it activates Switch 2, which opens the normally closed *SW 2-1* contacts. This action breaks the *DIRECTION* relay seal, allowing it to become de-energized, and the motor switches direction—now going back toward the bottom.

Rung 3 controls the *GRIPPER* solenoid. Controlling the gripper is simple because the *GRIPPER* needs to be activated for the same time period as the *DIRECTION* relay is energized—that is, from the time when the motor starts its upward journey, until just before it descends.

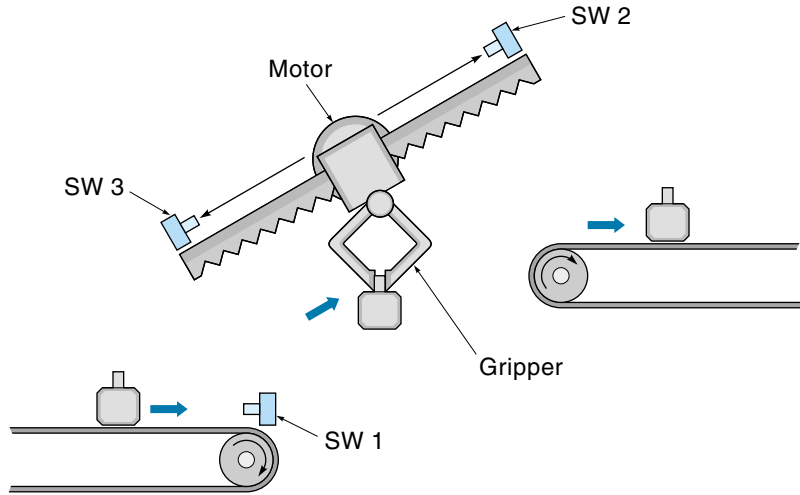
The final action in the cycle is when the gripper descends (empty) and activates Switch 3. This action opens the normally closed contacts (*SW 3-1*), breaking the seal in rung 1 (which had held the *MOTOR* relay energized), and the motor stops.

**Timers, Counters, and Sequencers**

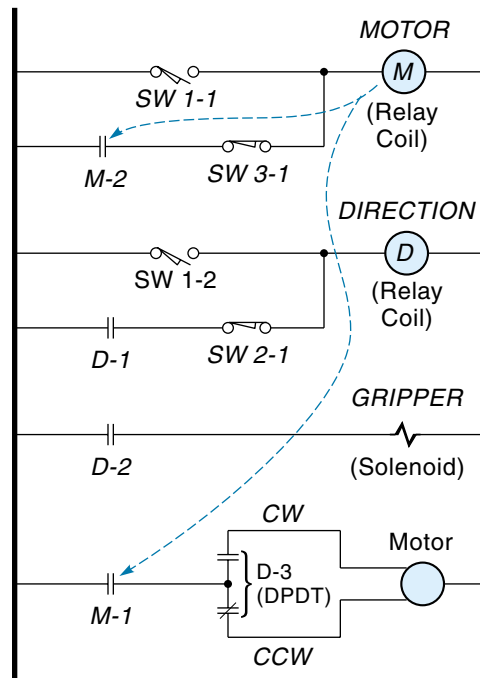
Many control situations require that a time delay be inserted at some point in the process. For example, a mixing operation might take 90 s, or a conveyer belt might be allowed 10 s to reach speed before parts are placed on it. A relay control system would use a **time-delay relay** to create the time delay. Once activated, the time-delay relay will wait a predetermined period of time before the contacts open (or close)—the delay

**Figure 12.6**

Pick-and-place robot control  
(Example 12.2).



(a) Hardware



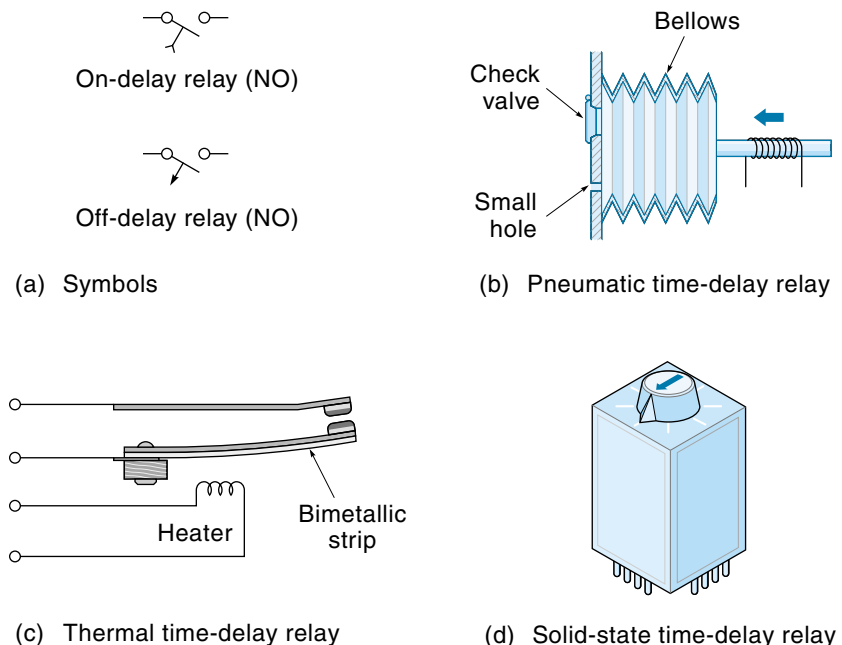
(b) Ladder diagram

may range from less than a second to minutes. Time-delay relays are categorized as being either *on-delay* or *off-delay* and are best explained by example. A 5-s **on-delay relay** with NO contacts will wait 5 s (after being energized) before closing its contacts. When the relay is de-energized, the contacts will open immediately. An application of the on-delay function is a security system that delays activation for a period of time after being turned on, to allow the occupants time to leave. The **off-delay relay** provides a time delay when the coil is de-energized. For example, a 5-s off-delay relay with NO contacts would close its contacts immediately when energized; when it is deenergized, however, the contacts would remain closed for 5 more seconds before opening. An application of the off-delay function is car lights that remain on for a period of time after being turned off (to provide light for the occupants as they are leaving the vehicle). Figure 12.7(a) shows the symbols used for time-delay relays.

Several different designs of delay timers, which use different operating principles, have evolved over the years. The **pneumatic time-delay relay** [Figure 12.7(b)] works as follows: When the relay is activated, a small spring-loaded bellows is squeezed closed, causing the air to escape through a check valve. The bellows then is allowed to slowly expand (the air being admitted through a small hole). When the bellows reaches its normal size again, the contacts close. These relays are available with a fixed or an adjustable delay.

A **thermal time-delay relay** [Figure 12.7(c)] uses a temperature-sensitive bimetallic strip. When the relay is energized, a small resistance heater warms the bimetallic

**Figure 12.7**  
Time-delay relays.





strip. As the strip heats, it bends and eventually closes the contacts. The flasher unit in a car, which controls the flashing directional signals, works on this principle.

**Solid-state time-delay relays** are used in most newer systems that require delay relays. An example is shown in Figure 12.7(d); the knob is for setting the delay. These units are based on the delay involved when (1) charging a capacitor or (2) counting high-speed clock pulses with a digital counter: the higher the count, the longer the delay.

### EXAMPLE 12.3

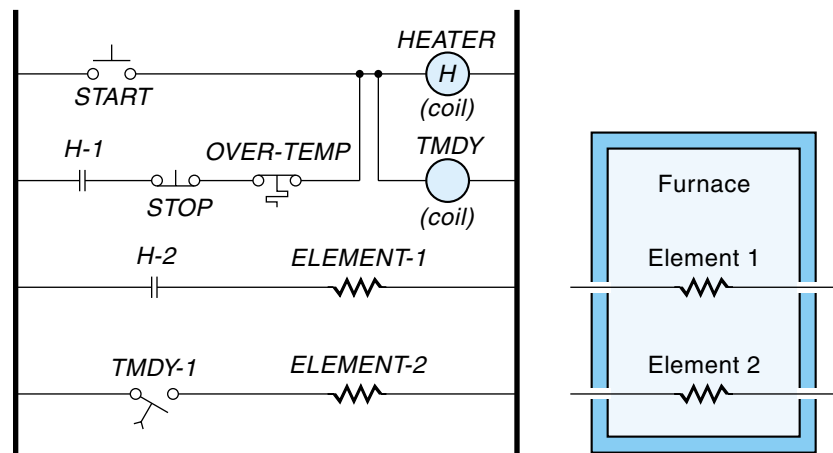
A small electric furnace has two heating elements that are energized in stages 3 min apart. That is, when the furnace is turned on, the first heating element comes on right away, and the second element comes on 3 min later. A temperature sensor will shut down the furnace if it gets too hot. Draw the ladder diagram for the control circuit.

### SOLUTION

On the top rung of the ladder diagram shown in Figure 12.8, a push-button switch energizes the *HEATER* relay, and the relay is sealed by the *H-1* contacts. This action supplies power to the first heating element (*ELEMENT-1*), via the relay contacts *H-2*, and energizes the *TMDY* delay relay (on-delay type). After 3 min, the time-delay relay activates the second heating element via the *TMDY-1* contacts (bottom rung). At any time if the stop button is pushed or the *OVER-TEMP* sensor contacts open, the seal is broken, and the *HEATER* relay deenergizes, causing both heating elements to shut down.

**Figure 12.8**

Furnace heater control  
(Example 12.3).

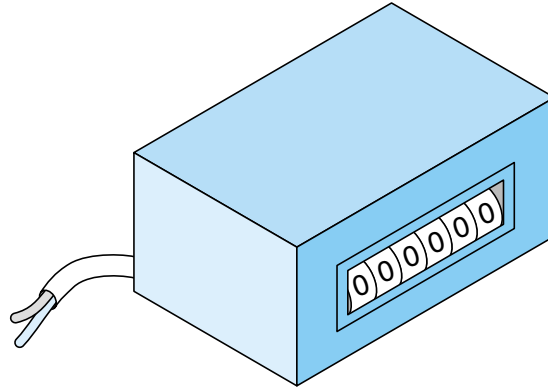


(a) Ladder diagram

(b) Furnace

**Figure 12.9**

An electromechanical counter.

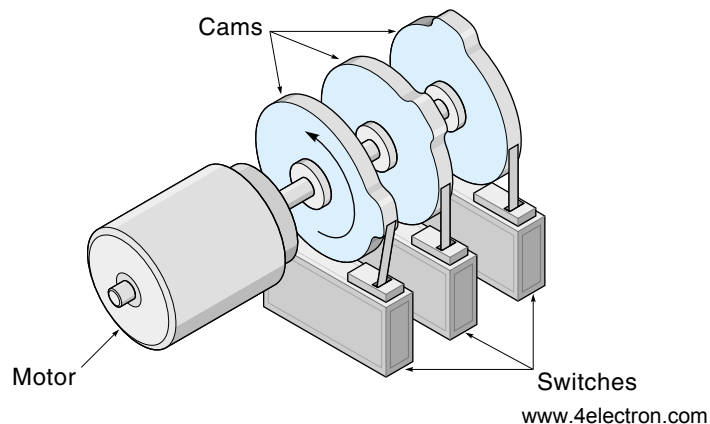


**Electromechanical counters** are devices that count events. The counter, which has a readable output similar to a car's odometer (Figure 12.9), increments once for each electric pulse received. Counters usually simply keep track of the number of times an operation is performed. For example, it could count the number of products coming down the assembly line. Some models will close a set of switch contacts when the count gets to a preset value.

**Electromechanical sequencers** control the process that has a sequence of timed operations. An example of this is a dishwasher controller. The water is admitted for so many seconds, then the wash cycle is activated for so many seconds, and then the water is pumped out for so many seconds, and so on. As shown in Figure 12.10, the sequencer (known as a *drum controller*) consists of a small timing motor that slowly rotates a cluster of cams, and the cams activate switches. Each switch controls one of the timed operations, and the whole sequence repeats itself with each revolution of the drum. Some sequencers use a solenoid ratchet mechanism to rotate the cams in small discrete steps—one step for each input pulse.

**Figure 12.10**

An electromechanical sequencer.



## 12.2 PROGRAMMABLE LOGIC CONTROLLERS

### Introduction

A **programmable logic controller** (PLC) is a small, self-contained, rugged computer designed to control processes and events in an industrial environment—that is, to take over the job previously done with relay logic controllers. Figure 12.11 shows a number of different types of PLCs. Physically, they range in size from that of a camera to that of a shoe box (and sometimes larger). Wires from switches, sensors, and other input devices are attached directly to the PLC; wires driving lights, small motors, motor starters, and other output devices are also connected directly to the PLC (Figure 12.12). Each PLC contains a microprocessor that has been programmed to drive the output terminals in a specified manner, based on the signals from the input terminals. The PLC program is usually developed on a separate computer such as a personal computer (PC), using special software provided by the PLC manufacturer. Once the program has been written, it is transferred, or downloaded, into the PLC. From this point on, the PLC may operate on its own, as a completely independent controller.

### PLC Hardware

Figure 12.13 shows the block diagram for a PLC and includes the fundamental parts found in any microprocessor system (as discussed in Chapter 2). These major functional blocks are explained next.

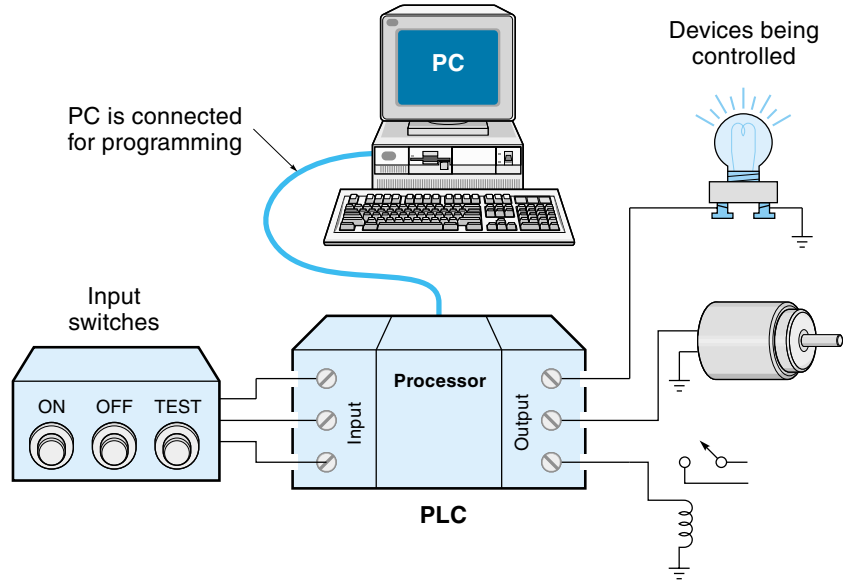
**Figure 12.11**

A selection of PLCs.  
(Allen-Bradley products  
courtesy of Rockwell  
Automation)



**Figure 12.12**

A PLC and its related components.



**Power Supply**

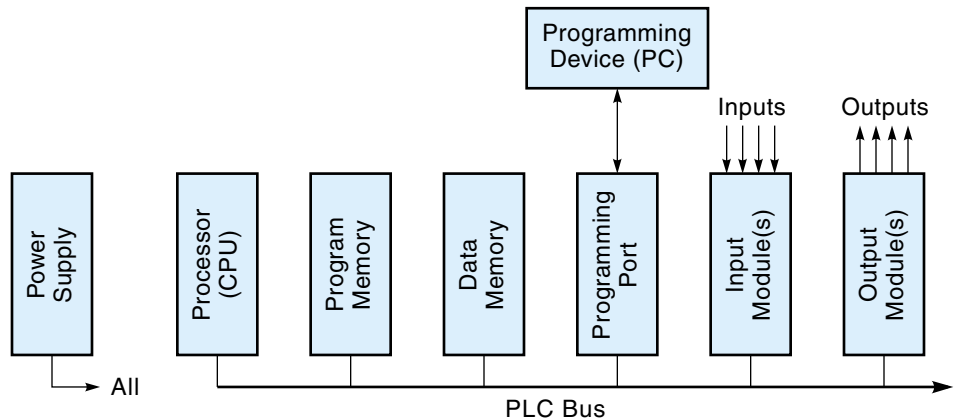
PLCs are usually powered directly from 120 or 240 Vac. The power supply converts the AC into DC voltages for the internal microprocessor components. It may also provide the user with a source of reduced voltage to drive switches, small relays, indicator lamps, and the like.

**Processor**

The **processor** is a microprocessor-based CPU and is the part of the PLC that is capable of reading and executing the program instructions, one-by-one (such as the rungs of a ladder logic program).

**Figure 12.13**

Block diagram of a PLC.



### *Program Memory*

The **program memory** receives and holds the downloaded program instructions from the programming device. If this memory is standard RAM, the program will be lost every time the power is turned off, requiring it to be reloaded. To avoid this bother, the program memory may use an EEPROM (electrically erasable programmable ROM) or a battery-backup RAM, both of which are capable of retaining data even when the power is off. An EPROM (UV erasable PROM) can also store the program, but this device requires a special programming unit that is not part of the PLC system.

### *Data Memory*

**Data memory** is RAM memory used as a “scratch pad” by the processor to temporarily store internal and external program-generated data. For example, it would store the present status of all switches connected to the input terminals and the value of internal counters and timers.

### *Programming Port*

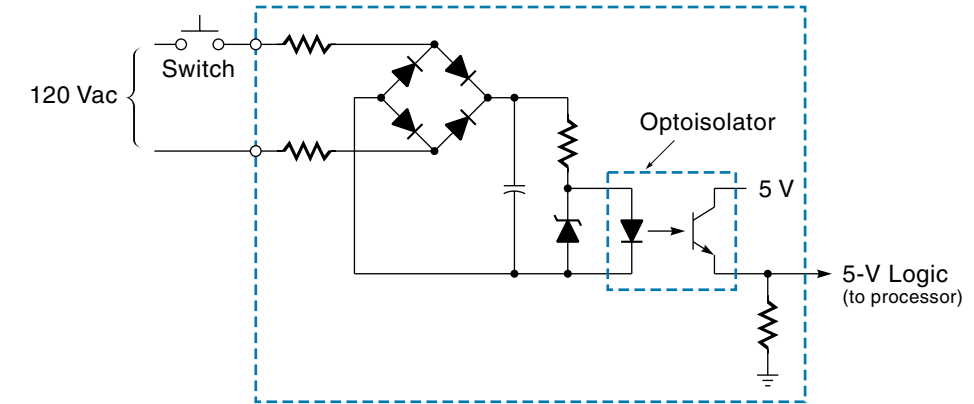
The **programming port**, an input/output (I/O) port, receives the downloaded program from the programming device (usually a PC). Remember that the PLC does not have an elaborate front panel or a built-in monitor; thus, to “see” what the PLC is doing inside (for debugging or troubleshooting), you must connect it to a PC, as illustrated in Figure 12.12.

### *Input and Output Modules*

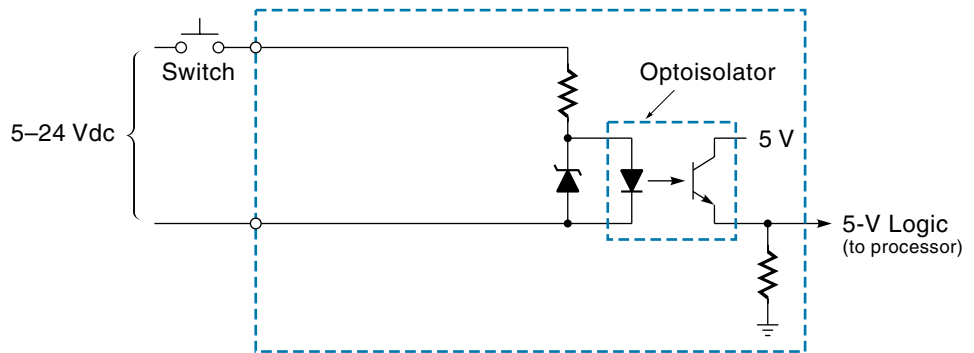
The **I/O modules** are interfaces to the outside world. These control ports may be built into the PLC unit or, more typically, are packaged as separate **plug-in modules**, where each module contains a set of ports (see Figures 12.11 and 12.12). The most common type of I/O is called **discrete I/O** and deals with on-off devices. *Discrete input modules* connect real-world switches to the PLC and are available for either AC or DC voltages (typically, 240 Vac, 120 Vac, 24 Vdc, and 5 Vdc). As shown in Figure 12.14, circuitry within the module converts the switched voltage into a logic voltage for the processor. Notice that an AC switch voltage must first be converted into a DC voltage with a rectifier. Also, an input module will usually include an opto-isolator to prevent possible high-voltage spikes from entering the PLC. A typical discrete input module would have 4, 8, or 16 inputs.

*Discrete output modules* provide on-off signals to drive lamps, relays, small motors, motor starters, and other devices. As shown in Figure 12.15, several types of output ports are available: Triac outputs control AC devices, transistor switches control DC devices, and relays control AC or DC devices (and provide isolation as well). A typical discrete output module would have 4, 8, or 16 outputs.

**Figure 12.14**  
PLC input  
module circuits.



(a) AC input module circuit



(b) DC input module circuit

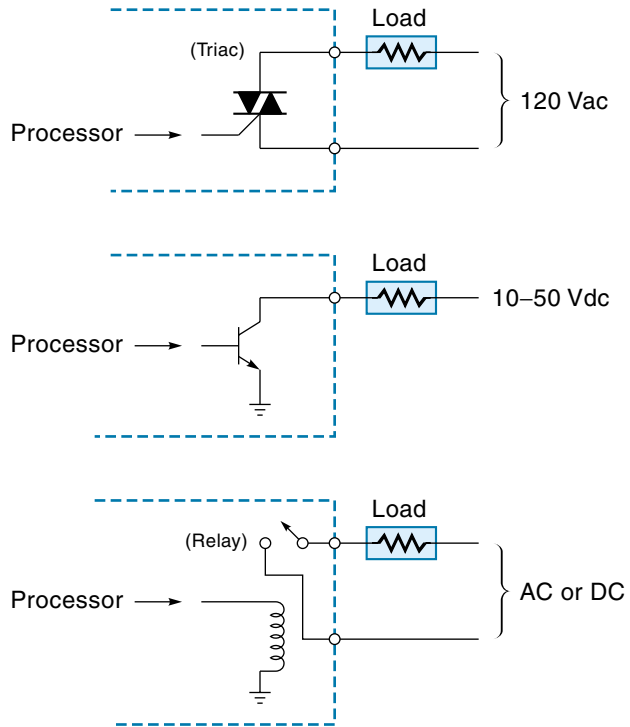
**Analog I/O modules** allow the PLC to handle analog signals. An *analog input module* [Figure 12.16(a)] has one or more ADCs (analog-to-digital converters), allowing analog sensors, such as temperature, to be connected directly to the PLC. Depending on the module, the analog voltage or current is converted into an 8-, 12-, or 16-bit digital word. An *analog output module* [Figure 12.16(b)] contains one or more DACs (digital-to-analog converters), allowing the PLC to provide an analog output—for example, to drive a DC motor at various voltage levels.

Specialized modules that perform particular functions are available for many PLCs. Examples include :

- *Thermocouple module*—Interfaces a thermocouple to the PLC.
- *Motion-control module*—Runs independently to control multi-axis motion in a device such as a robot (covered later in this chapter).
- *Communication module*—Connects the PLC to a network.

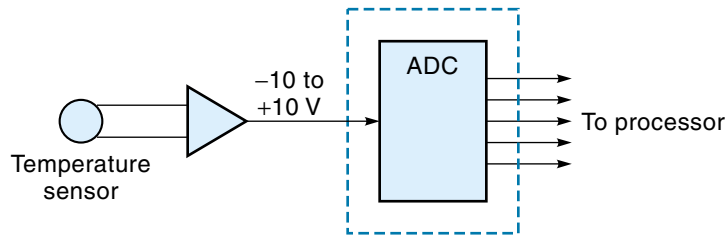
**Figure 12.15**

PLC output module circuits.

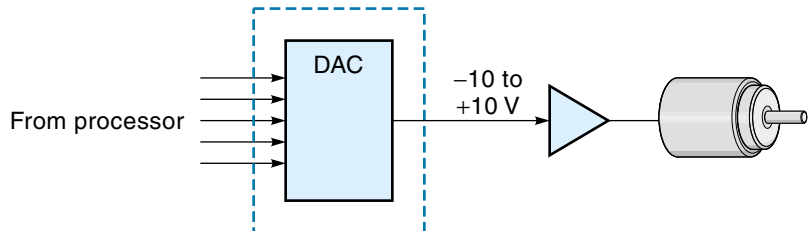


**Figure 12.16**

Analog I/O modules.



(a) Analog input module



(b) Analog output module

- *High-speed counter module*— Counts the number of input pulses for a fixed period of time.
- *PID module*— An independently running PID self-contained controller (PID control can also be implemented with software, as described later in this chapter).

### PLC Bus

The **PLC bus** (Figure 12.13) are the wires in the *backplane* of a PLC modular system, which contains the data bus, address bus, and control signals. The processor uses the bus to communicate with the modules.

### PLC Setup Procedure

The PLC evolved as a replacement for hard-wired relay logic and was designed to be installed, programmed, used, and maintained by technicians who were familiar with industrial relay and switching circuits. For this reason, manufacturers of PLCs developed a programming language that allows the user to program the desired control logic in the form of a ladder diagram. The ladder diagram may be entered into the system by either drawing it on the PC screen or using a specially adapted switch box (called a *hand-held programmer* (HHP)). If the PC is used, special software must be installed before the ladder diagram can be entered.

Assuming that the ladder diagram has been entered into a PC (or PC-like terminal device), the next step is to connect the PC to the PLC. In most cases, simply connect the two units with a cable or an interface circuit. A more sophisticated system may have a number of PLCs connected to a single PC on a LAN (local area network). With this system, the PC can communicate with any PLC on the network. Once connection has been established, the program is downloaded from the PC into the PLC.

With the program loaded, the PLC is ready to operate, and in any real application, the next step is testing and debugging. Testing can be done one of two ways: the total software approach, where the inputs are simulated from the programming terminal (PC), or a more real-world approach, where the inputs and outputs are connected to their actual destinations. In either case, the PC screen becomes a “window” into the workings of the program; if a problem becomes apparent, the program can be corrected immediately and testing continued until all bugs are worked out. When the testing is complete, the PC can be disconnected, and the PLC will continue to execute its stored program as an independent unit.

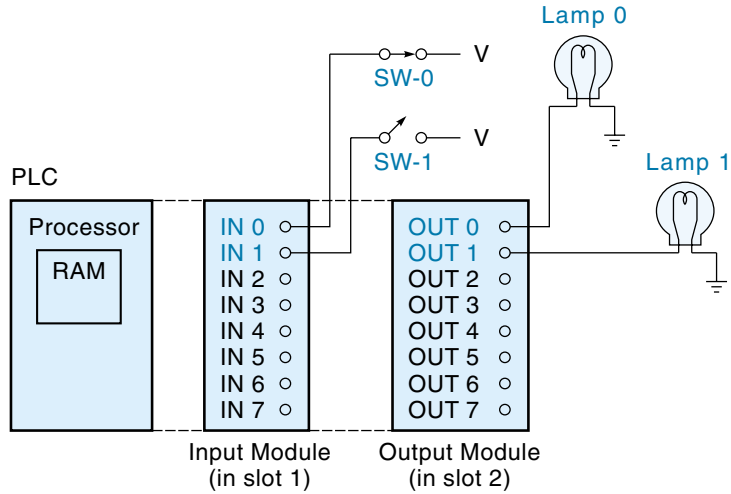
### PLC Operation

The PLC accomplishes its job of implementing the ladder logic control program in typical computer fashion—one step at a time. To see how it does this, let’s examine a simple PLC application of turning lamps on and off [Figure 12.17(a)]. Notice that the PLC has one input module and one output module. Two external switches (SW-0 and SW-1) are connected to the PLC via terminals *IN-0* and *IN-1* of the input module. Two termi-

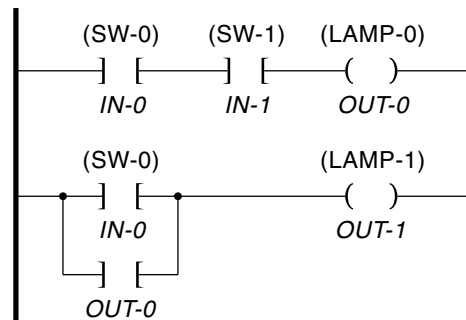


**Figure 12.17**

A simple PLC application and ladder diagram.



(a) Hardware



(b) Ladder diagram

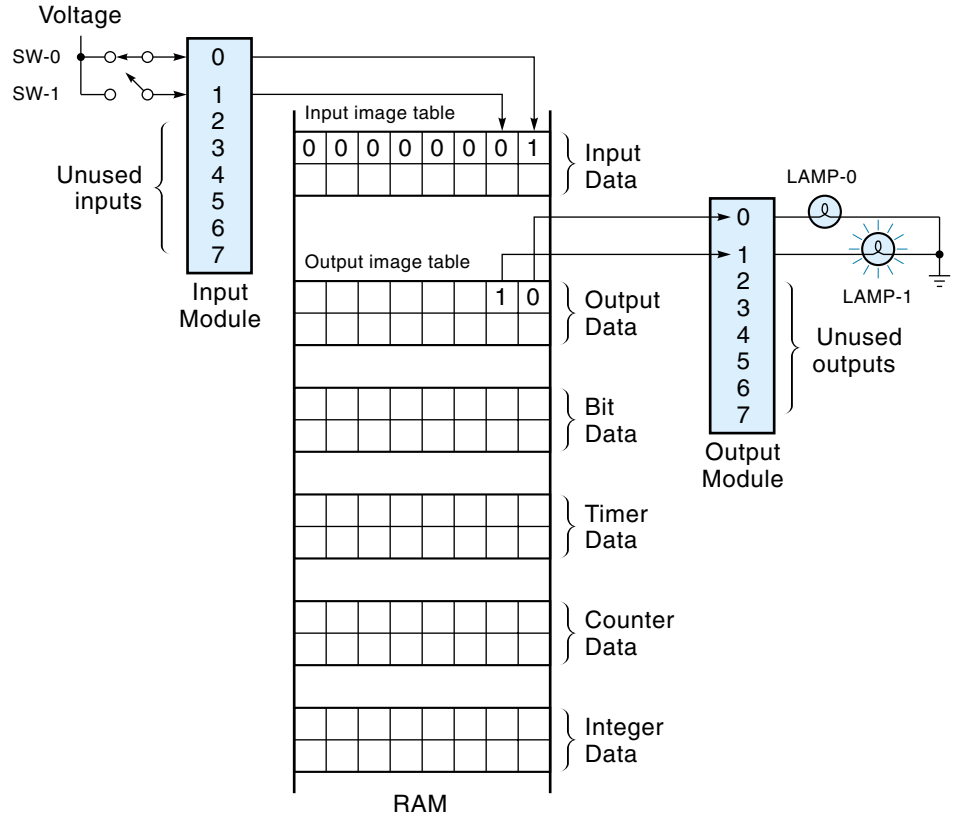
nals of the output module (*OUT-0* and *OUT-1*) drive two indicator lamps (*LAMP-0* and *LAMP-1*). The ladder diagram for this application has only two rungs [Figure 12.17(b)]. The top rung will light *LAMP-0* if *both* *SW-0* and *SW-1* are closed. The bottom rung will light *LAMP-1* if *either* *SW-0* or *OUT-0* are closed. (The *OUT-0* contacts can be thought of as NO relay contacts of coil *OUT-0* in rung 1.)

Assume that the program has been loaded into the PLC and is residing in the program section of memory. When the PLC is set to **run mode**, program execution begins. Execution is accomplished as a series of steps, which are repeated over and over. One complete cycle of all these steps is a **scan**. The steps are as follows:

1. The PLC reads in data from the input module and stores the values of all eight inputs, as a word, in the Input Data section of memory (Figure 12.18). Each bit in the word represents the present status of one external switch.

**Figure 12.18**

Data in memory, showing the relationship between memory position and the I/O terminals.



- The PLC turns its attention to the ladder diagram, starting with the first rung. The first two symbols in the rung are input switches, so the processor reads their status *from* RAM. It then evaluates the logic indicated by the rung. In this case, SW-0 is on, and SW-1 is off, so *OUT-0* will *not* be energized. This status of *OUT-0* is stored as a 0 bit in the Output Data section of RAM (Figure 12.18).
- After completely dealing with the first rung, the PLC turns its attention to the second rung. The status of *IN-0* and *OUT-0* are read from RAM. Using these data, the PLC performs the rung logic, which is an OR operation between *IN-0* and *OUT-0* (recall that *OUT-0* is the output of the top rung and is presently off). In this case, because SW-0 is on, *OUT-1* will be energized, and this result is stored as a 1 bit in the Output Data section of RAM. This completes action on the second rung. If there were more rungs, each would be evaluated in sequence until all had been executed.
- As a final action of the scan, the updated output data in RAM are sent to the output module, causing all eight output terminals to be updated at once. In this case, LAMP-0 would be off, and LAMP-1 would be on. Now the PLC will go into a rest state for a few seconds or less until the designated scan time has elapsed, and then the whole process starts over again with step 1.

Looking over the process just described, some observations are in order. First, the PLC accomplishes its control mission in the same way that all digital controllers work—by executing a loop (scan) over and over again. (For the PLC, however, the looping action is automatic and does not have to be programmed.) For each scan, inputs are read in, outputs are calculated based on the inputs, and then the outputs are sent out to the real world. This means that as much as one scan time's delay may be between an input and an output. Such a delay is not present in a true hard-wired relay circuit where the only delay would be the propagation time of the relays. This leads to the second observation: The order of the rungs in the ladder diagram may be important because the rungs are executed sequentially from top to bottom.

A final observation of the PLC hardware operation is that, for all practical purposes, it can do many independent control operations at the same time. We may be conditioned to think of a computer as being able to do only one thing at a time, albeit very rapidly. And, as we have seen, the PLC does execute its program one rung at a time. However, each rung is a separate circuit, and each control application (as far as the PLC is concerned) is only a set of rungs. Therefore, you could have one PLC running three different control applications at the same time. Each application would have its own assigned I/O terminal and its own set of rungs in the program. Within one scan cycle of the program, the PLC would update the three sets of outputs. This is similar to the chess master who plays three opponents simultaneously by going down the line, making one move on each board.

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## 12.3 PROGRAMMING THE PLC

### Ladder Diagram Programming

The most common way to program the PLC is to design the desired control circuit in the form of a relay logic ladder diagram and then enter this ladder diagram into a programming terminal (which could be either a PC, a dedicated PLC programming terminal, or a special handheld switch box). The programming terminal is capable of converting the ladder diagram into digital codes and then sending this program to the PLC where it is stored in memory. There are other ways to program a PLC besides ladder diagrams, and these will be introduced later in this chapter.

There are currently many manufacturers of PLCs, and though they all have basic similarities, the details of the hardware and software will, of course, be different. The following discussion on programming will be kept as general as possible but is patterned after the Allen-Bradley line of PLCs. This seems a logical choice because Allen-Bradley is one of the pioneers in the PLC field and its products are respected and well used by industry.

The PLC deals with two kinds of data that are maintained in memory: a **program file** in RAM or EEPROM, which contains the ladder logic program, and **data files** in

RAM, which hold the changing data from the control application, such as switch settings. Table 12.1 lists some of the more common types of data files. Each data file occupies a portion of RAM memory and consists of any number of 8- or 16-bit words. This data arrangement is illustrated in Figure 12.18.

The Input and Output Data files are handled in a special way. The PLC automatically transfers the on-off condition of all input switches directly to the Input Data file in RAM as part of the scan. This is called making an *input image table* (as indicated in Figure 12.18). Notice that a closed switch becomes a 1 in RAM and an open switch becomes a 0. Similarly, data in the Output Data file (called the *output image table*) are automatically transferred to the output-module terminals.

Thus, the PLC program actually deals with only the image data in their various data files. That is why each device on the ladder diagram—be it switch, relay, timer, or whatever—is identified by its address in RAM. Even though the programmer may think of a certain switch as being, say, an overflow limit switch, the software thinks of it as being a particular bit in a particular data file.

There is an important fundamental difference between a relay logic ladder diagram and a PLC ladder diagram. *The relay ladder diagram is basically a wiring diagram*, and a rung becomes active when current can flow from one power rail to the other; that is, if there is a path for the current through the contacts, then the load will be energized. On the other hand, *the PLC ladder diagram is basically a logic diagram* consisting of symbols (called input and output instructions), and a rung becomes continuous when there is a logical TRUE path from rail to rail. Referring to Figure 12.17(b), if there is a continuous path of TRUE Input instructions on the left side of a rung, then the Output instruction (on the right side) will become TRUE. Therefore, writing a PLC program involves placing instructions in rungs so as to obtain the desired result. Naturally, you can only use instructions that are supported by the language being used, but most PLCs support the basic components such as switches, relays, timers, counters, and sequencers. We will discuss each type of instruction next.

**TABLE 12.1**

**Types of Data Files**

	Data file	Description
I	Input Data	The current status of externally connected switches
O	Output Data	The status of those devices specified as outputs
B	Bit Data	The “scratch pad” to store individual bits
T	Timer Data	The data associated with timers
C	Counter Data	The data associated with counters
R	Control Data	The data associated with sequencers and other devices
N	Integer Data	Numbers, such as temperatures (in binary form)

## Bit Instructions

Switches and relays are referred to as **bit instructions** because it takes only 1 bit to describe if a switch is open or closed. There are two kinds of switch instructions: one represents a NO switch and the other a NC switch. The following are the symbol and instruction for the NO switch:

NO instruction: —] [— (EXAMINE IF ON, Allen-Bradley)

When this symbol is placed in the ladder diagram, it behaves as a NO switch, push button, or set of relay contacts. It can be activated either by a real-world switch or another logical switch on the ladder diagram. In either case, when activated, the NO instruction in the ladder diagram goes TRUE. The state of the NO instruction is maintained by a particular bit in memory, where a 0 means off (FALSE) and a 1 means on (TRUE). The address of this bit is placed on the ladder diagram next to the instruction symbol.

The following are the symbol and instruction for the NC switch:

NC instruction: —]/[— (EXAMINE IF OFF, Allen-Bradley)

When this symbol is placed in the ladder diagram, it behaves logically in the same way any NC switch does—that is, it is normally TRUE and when activated goes FALSE (it behaves as a logical inverter). Care must be taken when using this instruction in conjunction with a real-world NC switch because it may not behave as you think. The problem is that people lose sight of the fact that a real-world NC switch and the NC instruction are *not* the same entity but, rather, *one drives the other*; that is, the physical switch controls the state of the NC instruction. Suppose you actually wired a real NC switch to the PLC and used it to control an NC instruction. Then, if the real switch was activated, it would send a 0 to the PLC, and the logical NC instruction would invert this FALSE to a TRUE, which may be backward from what you expected. To avoid this pitfall, remember that the real-world switch may drive the logical switch, but they are *not* the same thing.

The following symbol represents a relay coil or simply an output signal:

OUTPUT —( )— (OUTPUT ENERGIZE, Allen-Bradley)

If there is a continuous TRUE path through the rung, then the OUTPUT will go TRUE; thus, if this symbol represents a relay coil, the relay will energize. What really happens inside the PLC is that the bit in memory corresponding to this instruction goes to a 1. This bit may be used to drive an actual hardware relay through an output module or to control an instruction in another rung, or both. In fact, there is no limit to the number of times this bit can be used by other instructions in the program (which is another advantage PLCs have over hard-wired relay logic).

The operation of the three bit instructions are summarized in Table 12.2.

TABLE 12.2

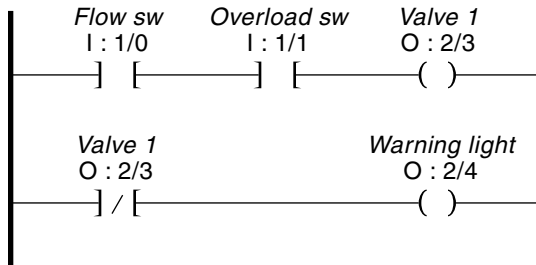
## Summary of Bit Instructions

If the data file bit is	NO instruction —] [—	NC instruction —]/[—	OUTPUT —( )—
Logic 0	FALSE	TRUE	FALSE
Logic 1	TRUE	FALSE	TRUE

In a PLC ladder diagram, each symbol is accompanied by its RAM address (Figure 12.19) because this address tells the PLC where to find the present logical state of the symbol. Recall that the logical state of each symbol is always maintained in RAM. Titles of real-world components that the symbols represent, such as *Overload Sw*, are frequently added to improve clarity, but the PLC itself deals strictly with addresses. The address of an individual bit may be specified by three quantities: the file type, word number, and position of the bit within the word.\* As previously stated, I/O signals usually have a direct relationship between their address and the physical connection point on an I/O module. For example, in the top rung of Figure 12.19, the address of *Flow sw* is given as I : 1/0. The I stands for Input Data file, the 1 means that the input module is plugged into slot 1 of the PLC chassis, and the 0 means that the switch is wired to terminal 0 of the module. The following is a more general interpretation of the *Flow sw* instruction address:

Type of data file (Input in this case)  
 Address of word within the data file (or I/O module slot #)  
 Bit position within the word, if appropriate (or terminal # of I/O module)  
 I:1/0  
 ———] [—————

**Figure 12.19**  
 A simple PLC  
 program.



\*The address system used in this chapter is a simplified version of the Allen-Bradley system.

The second NO instruction on the first rung (Figure 12.19) represents a switch wired to terminal 1 of the same input module used by *Flow sw*. The OUTPUT instruction activates terminal 3 of the output module plugged into slot 2.

The operation of the top rung in Figure 12.19 is as follows: The *Valve 1* output will go TRUE when there is continuity through the rung. Continuity will be established when both NO instructions are TRUE—in other words, when the contacts of both *Flow sw* and *Overload sw* are closed.

Operation of the second rung illustrates two important concepts. First, notice that the address of the NC instruction is the same as the OUTPUT instruction in the top rung, meaning that the NC instruction will be controlled by the *Valve 1* bit (located at address 0:2/3). Second, because it is an NC instruction, it will go TRUE only when *Valve 1* is FALSE. So *Warning light* will come on when the valve goes off.

## Timers

The Timer instruction provides a time delay, performing the function of a time-delay relay. Examples of using time delays include controlling the time for a mixing operation or the duration of a warning beep. The length of time delay is determined by specifying a preset value. The timer is enabled when the rung conditions become TRUE. Once enabled, it automatically counts up until it reaches the Preset value and then goes TRUE (and stays TRUE). The symbol for the Timer instruction is a box or a circle with the pertinent information placed in or about it, as shown and explained in Figure 12.20. Notice that the Timer instruction is placed in the rung. When *Switch I : 1/3* goes TRUE, the timer starts counting. Five seconds later, when the Accumulator value reaches the Preset value, the DN (done) bit goes TRUE. Notice that the NO instruction in the second rung has the DN bit from Timer T : 1 as its address. Therefore, it will go true after 5 s, causing the OUTPUT instruction at 0 : 2/4 to turn on a light.

In Figure 12.20, the timer is acting as an on-delay relay because it starts the delay when the rung goes TRUE. Off-delay timer instructions are also available.

### EXAMPLE 12.4

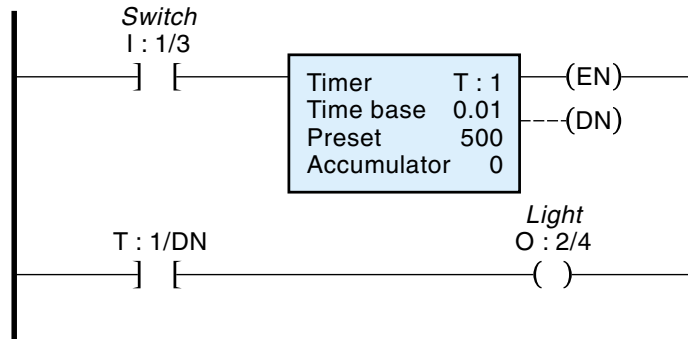
A batch process—which involves filling a vat with a liquid, mixing the liquid, and draining the vat—is automated with a PLC. Figure 12.21(a) shows the hardware. The specific sequence of events is as follows: When the start button near the process is pushed:

1. A fill valve opens and lets a liquid into a vat until it's full.
2. The liquid in the vat is mixed for 3 min.
3. A drain valve opens and drains the tank.

Draw the ladder diagram for the PLC program.

**Figure 12.20**

The timer instruction.



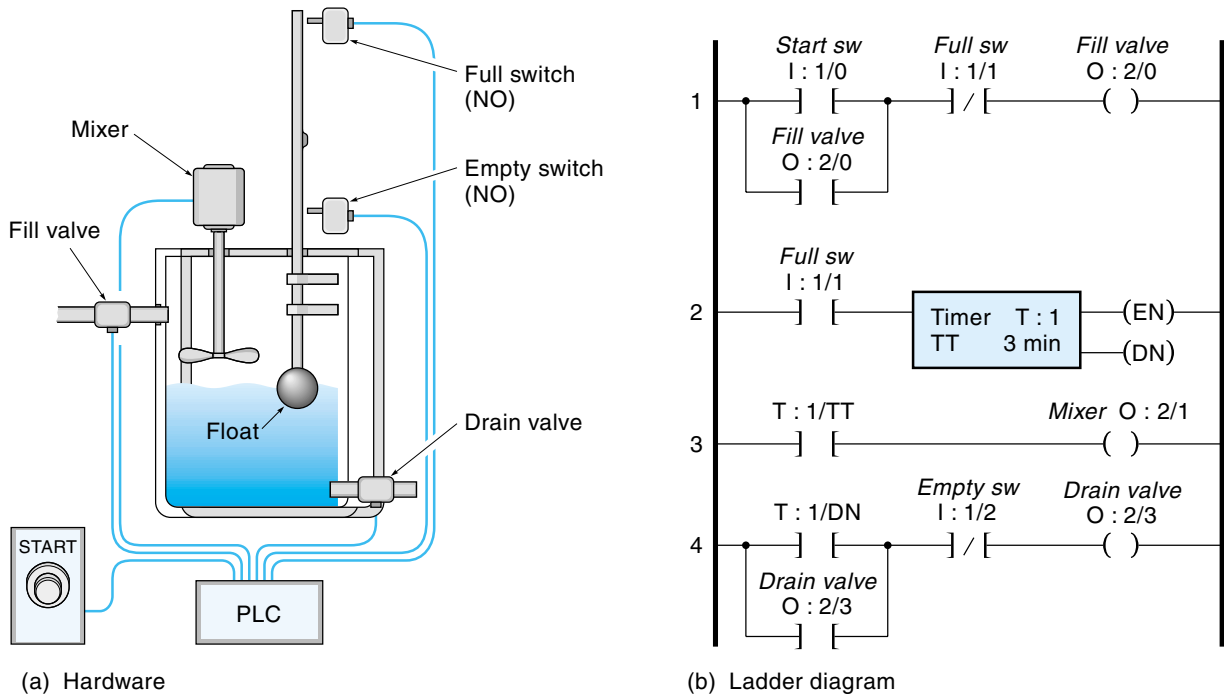
T : 1	The timer address (actually the first of three addresses in RAM) where the first address holds the status bits EN, TT, and DN; the second address holds the preset value; and the third address holds the accumulator value.
Time base	The value of 0.01 means that each count corresponds to 0.01 seconds.
Preset	The value of 500 means that the delay will last 500 counts, which in this case is 5 s ( $0.01\text{s} \times 500 = 5\text{ s}$ ).
Accumulator	Holds the value of the current count.
EN	A bit that is TRUE as long as the timer rung is TRUE.
TT	A bit that is TRUE as long as the timer is counting, that is, is TRUE for the time delay.
DN	A bit that goes TRUE when the timer is done—in other words, when the count gets to the preset value.

**SOLUTION**

Figure 12.21(b) shows the completed ladder diagram, which is explained rung-by-rung:

1. Rung 1 activates the fill valve and will go TRUE (and seal) when the local-control start switch is momentarily activated *and* the tank is less than full. The physical float-activated NO switch (Full switch) drives the *Full sw* NC instruction in rung 1; therefore, the NC instruction will be TRUE as long as the vat is not full (it changes to FALSE when the vat is full). As long as the entire rung is TRUE, the OUTPUT instruction (*Fill valve*) will be TRUE, which turns on the actual fill valve. (During the fill period, none of the other rungs are TRUE.)



**Figure 12.21**

Using a PLC timer instruction to control a batch-mixing operation.

2. Rung 2 activates the 3-min mixing timer. When the vat is full, the physical full switch closes (goes to on), so the NO instruction (*Full sw*) goes TRUE, which starts Timer T : 1 (which has been programmed to provide the 3-min mixing time).
3. Rung 3 controls the mixer motor. As long as the TT bit of the timer is TRUE, the OUTPUT instruction *Mixer* will be TRUE, which activates the mixer motor.
4. Rung 4 activates the drain valve. After 3 min, the timer “times out,” and the DN bit (T : 1/DN) goes TRUE, making the first instruction in rung 4 go TRUE. The NC instruction (*Empty sw*) will also be TRUE because the vat is still full (that is, the Empty switch remains unactivated). Therefore, rung 4 will be continuous, causing the OUTPUT instruction *Drain valve* to go TRUE, which opens the drain valve, allowing the liquid to leave. Here’s where it gets interesting: As soon as the liquid starts to drain, the full switch will deactivate, making rung 2 go FALSE, which causes the timer’s DN bit to go FALSE. This latter action will cause the first instruction in rung 4 to go FALSE. So, to keep rung 4 TRUE until the vat has completely drained, the T : 1/DN instruction has a parallel branch instruction, which is activated by *Drain valve* itself. In effect, rung 4 is latched on until *Empty sw* activates and breaks the seal.

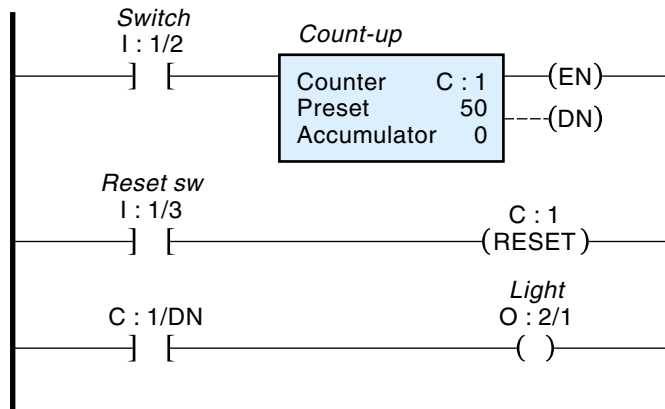
## Counters

A Counter instruction keeps track of the number of times some event occurs. The count could represent the number of parts to be loaded into a box or the number of times some operation is done in a day. Counters may be either count-up or count-down types. The Counter instruction is placed in a rung and will increment (or decrement) every time the rung makes a FALSE-to-TRUE transition. The count is retained until a RESET instruction (with the same address as the Counter) is enabled. The Counter has a Preset value associated with it. When the count gets up to the Preset value, the output goes TRUE. This allows the program to initiate some action based on a certain count; for example, after 50 items are loaded in a box, a new box is moved into place. The symbol for the Counter may be a box or a circle, with all pertinent data placed in or about it, as shown and explained in Figure 12.22.

The ladder diagram in Figure 12.22 is a simple circuit that counts the number of times that *Switch* (address I : 1/2) is opened and closed. The Counter can be reset at any

**Figure 12.22**

The Counter instruction.



C : 1	The counter address (actually the first of three addresses) where the first address holds such bits as CU and DN; the second address holds the preset value; and the third address holds the accumulated value.
Preset	When the counter gets to the Preset value, it makes the DN bit go TRUE and keeps on counting.
Accumulator	Holds the value of the current count.
EN	Goes TRUE when the counter rung is TRUE.
DN	Stands for <i>done</i> —goes TRUE when the count meets or exceeds the PRESET value.
RESET	A separate instruction with the same counter address; when this bit goes TRUE, the Accumulator resets to zero (resets the counter).

time by closing *Reset sw* because this action activates the RESET instruction. The third rung uses the DN bit from the Counter to turn on a light when the count gets to 50 or higher.

## Sequencers

The Sequencer instruction is used when a repeating sequence of outputs is required (recall the example of a dishwasher that cycles through predetermined steps). Traditionally, electromechanical sequencers (Figure 12.10) were used in this type of application (where a drum rotates slowly, and cams on the drum activate switches). The Sequencer instruction allows the PLC to implement this common control strategy.

Figure 12.23 shows the PLC Sequencer instruction. Operation is as follows: The desired output-bit patterns for each step are stored (sequentially) as words in memory—as usual, each bit in the word corresponds to a specific terminal of the output module. Every time the Sequencer instruction steps, it connects the next output pattern in memory to the designated output module. The address of the first pattern (word) in the sequence is given in the Sequencer instruction as the Sequencer File Adr (which is address B : 1 in Figure 12.23). The Sequencer shown in Figure 12.23 steps when the rung makes a FALSE-to-TRUE transition. Other versions of the Sequencer instruction allow the programmer to insert a time value for each step.

### EXAMPLE 12.5

A process for washing parts requires the following sequence:

1. Spray water and detergent for 2 min (*Wash cycle*).
2. Rinse with water spray only for 1 min (*Rinse cycle*).
3. Water off, air blow dry for 3 min (*Drying cycle*).

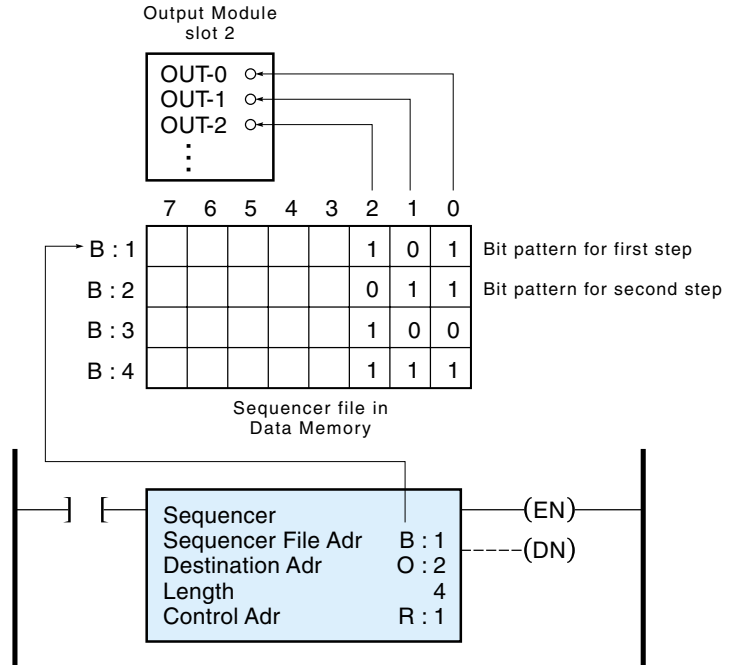
The sequence is started with a toggle switch. Draw the ladder diagram for this process (using the Sequencer instruction).

### SOLUTION

The completed ladder diagram (Figure 12.24) consists of six rungs. Rungs 1 and 2 generate the 2-min time delay for *Wash cycle*. Rungs 3 and 4 generate the 1-min time delay for *Rinse cycle*, and rung 5 generates the 3-min time delay for *Drying cycle*. Rung 6 has the Sequencer instruction. The Sequencer instruction specifies that the Sequence file starts at address B : 1, that the file is three words long, and that the output module is in slot 2. Looking at the Sequence file, we see that bit 0 controls the water valve, bit 1 controls the detergent valve, and bit 2 controls the blow-dry fan.

**Figure 12.23**

The Sequencer instruction.



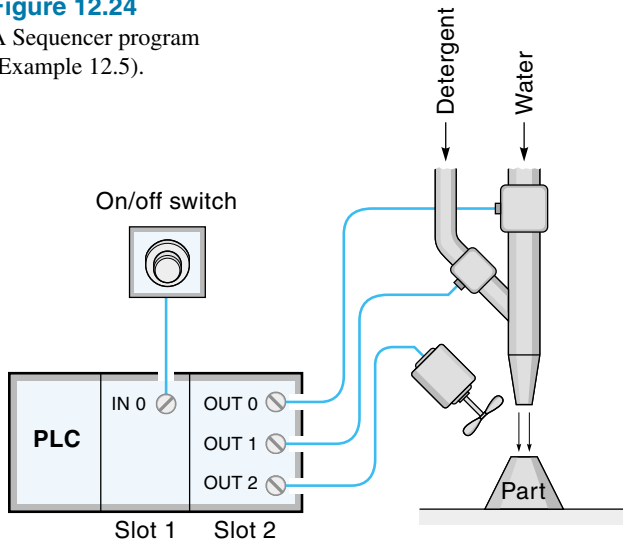
- Sequencer File Adr** The first address of the Sequence file, which stores the sets of outputs.
- Destination Adr** The output address (or slot) to which the Sequence file words are transferred with each step.
- Length** The length of the Sequence file, that is, the number of steps in the sequence.
- EN** Goes TRUE when the Sequencer rung is TRUE.
- DN** Stands for *done*—goes TRUE when it has operated on the last word in the Sequence file.
- Control Adr** Address that stores the control bits (EN, DN) and words (length) of the sequencer.

### Operation Overview

The program consists of five timer rungs and a sequencer rung. The timers are activated, one after the other, down the program—that is, each timer, when finished, starts the timer next in line. The outputs of Timers T : 1, T : 3, and T : 5 are used to step the Sequencer instruction. The Sequencer instruction presents three output-bit patterns to module 2.

**Figure 12.24**

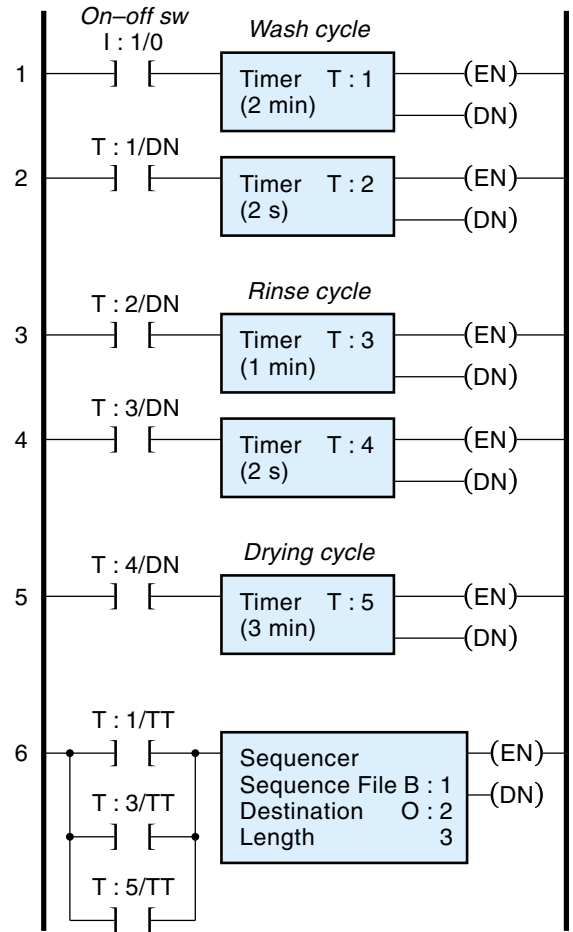
A Sequencer program  
(Example 12.5).



(a) Hardware

						(Blow-dry fan)	(Detergent valve)	(Water valve)
						2	1	0
B : 1						0	1	1
B : 2						0	0	1
B : 3						1	0	0

(b) Sequencer File



(c) Ladder diagram

**Detailed Operation**

Operation of the ladder program is as follows: The action starts on rung 1 when the operator switches the on-off toggle switch to on. This provides a FALSE-to-TRUE transition that starts Timer T : 1 (*Wash cycle*), causing its timing bit (TT) to go TRUE for 2 min. This same timing bit (T : 1/TT), makes rung 6 go TRUE, stepping the sequencer to its first position (connecting the word at address B : 1 to output module 0 : 2). Notice that bit 0 and bit 1 of word B : 1 are 1s, which cause terminals OUT-0 and OUT-1 to go on, turning on the water and detergent valves.

After 2 min, the DN bit of Timer T : 1 (T : 1/DN) goes TRUE, which starts the next Timer (T : 2 in rung 2). This timer is only 2 s long, and its purpose is to create a short gap between the steps (to allow the sequencer rung to reset).

When Timer T : 2 is done, its DN bit (T : 2/DN) starts the 1-min Timer T : 3 (*Rinse cycle*). Timer T : 3 timing bit (T : 3/TT) causes the Sequencer instruction to advance to the second step (B : 2). Notice that in word B : 2, only bit 0 is a 1, causing OUT-0 to remain on (leaving the water on, but turning off the detergent).

Following in the same manner, Timer T : 4 provides a short gap, and then Timer T : 5 advances the sequencer to the third and last position (B : 3) for the 3-min *Drying cycle*. Notice that bit 2 of word B : 3 becomes OUT-2, and turns on the blow-dry fan.

There are many possible ways to program this sequence. The method chosen (Figure 12.24) is not necessarily the shortest, but it is a clear, uncomplicated solution that in real life is more valuable than a clever short program (that no one can understand except the programmer!). Also, remember that extra rungs on a PLC ladder program do not mean more hardware (it does mean a slightly longer scan time and memory usage, which in most cases is not an issue).

## Using a PLC as a Two-Point Controller

Example 12.6 shows how a PLC can be used to implement a control strategy that requires decisions based on analog values. Although it is a simple example, it demonstrates the use of analog inputs and instructions that deal with analog values.

### EXAMPLE 12.6

The temperature in an electric oven is to be maintained by a 16-bit PLC at approximately 100°C, using two-point control (actual range: 98–102°). Figure 12.25(a) shows the system hardware: an oven with an electric heating element driven by a *contactor* (high-current relay), an LM35 temperature sensor (produces 10 mV/°C—see Chapter 6), an operator on-off switch, and the PLC. The PLC has a processor and three I/O modules: a discrete input module (slot 1), a 16-bit analog input module (slot 2), and a discrete output module (slot 3).

Draw the ladder diagram for this system.

### SOLUTION

We are asking the PLC to perform the job of a thermostat: When the oven temperature falls below 98°, the heating element will turn on and stay on until the

temperature reaches  $102^{\circ}$ . Thus, the PLC will be giving a discrete output (to the heater), based on an analog input (temperature).

The ladder diagram [Figure 12.25(b)] consists of three rungs that implement the control logic; however, recall that *before* the PLC executes the rungs, it reads in all input module data. In this case, it will read in the condition of the on-off switch from the input module in slot 1 (storing it to memory at address I : 1/0), and it will read in the value of the temperature through the analog input module in slot 2. The analog module contains an ADC that converts the analog voltage to a 16-bit word. The resolution of the module is given as

$$\text{ADC resolution} = 0.305176 \text{ mV} = 1 \text{ LSB}^*$$

Using this resolution, we can find the digital equivalent (in binary) of the lower- and upper-limit temperatures,  $98^{\circ}$  and  $102^{\circ}$ :

$$\text{Analog input voltage for } 98^{\circ} = 98^{\circ} \times \frac{10 \text{ mV}}{1^{\circ}\text{C}} = 980 \text{ mV}$$

}
  
Output of LM35

$$\text{ADC output for } 98^{\circ} = 980 \text{ mV} \times \frac{\text{LSB}}{0.305176 \text{ mV}} = 3211 \text{ LSB} = 110010001011$$

$$\text{Analog input voltage for } 102^{\circ} = 102^{\circ} \times \frac{10 \text{ mV}}{1^{\circ}\text{C}} = 1020 \text{ mV}$$

}
  
Output of LM35

$$\begin{aligned} \text{ADC output for } 102^{\circ} &= 1020 \text{ mV} \times \frac{\text{LSB}}{0.305176 \text{ mV}} = 3342 \text{ LSB} \\ &= 110100001110 \end{aligned}$$

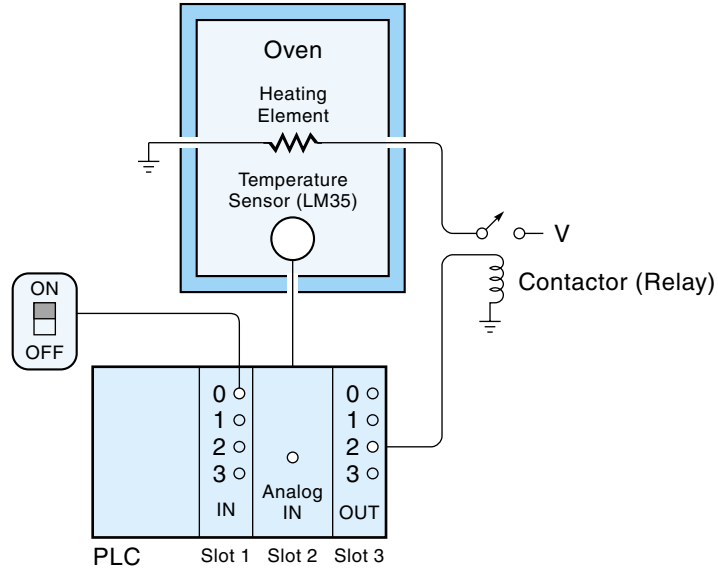
We now know the digital values of the upper- and lower-limit temperatures. We will need these numbers in the program. The operation of the ladder program follows [refer to Figure 12.25(b)]:

1. Rung 1 determines if the temperature is below  $98^{\circ}\text{C}$  and will go TRUE when the conditions of the Compare instruction are met. This particular Compare

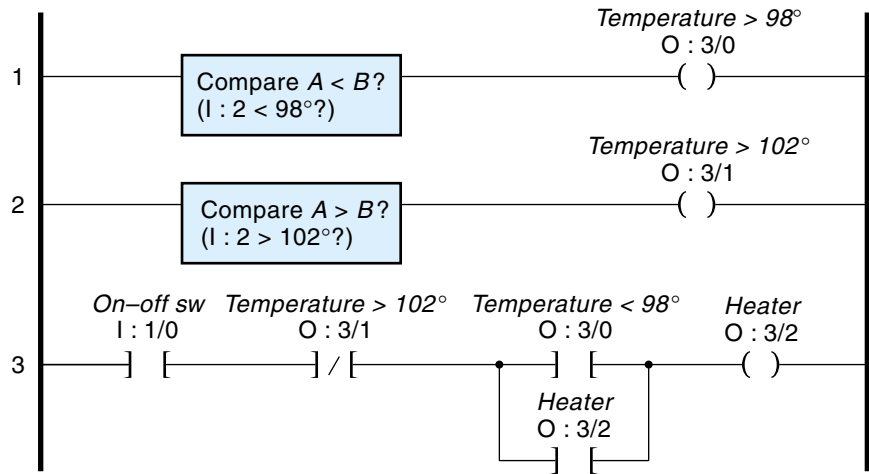
\*This ratio would be found on the specification sheet and comes from the following analysis: Input voltage range =  $\pm 10 \text{ V} = 20 \text{ V}$ ; output is 16 bits = 65,535 states (2s complement);  $\text{LSB} = 20 \text{ V}/65,535 = 0.305176 \text{ mV}$ .

**Figure 12.25**

A two-point controller program (Example 12.6).



(a) Hardware



(b) Ladder diagram



instruction compares two 16-bit words and goes TRUE if the value of word A is less than word B. In this case, word A is the actual temperature (as converted by the ADC and stored at address I : 2), and word B is the binary equivalent of the lower-limit temperature of 98°. Therefore, the rung will go TRUE if the oven temperature is less than 98°. The value of this rung is stored in bit 0 : 3/0.

2. Rung 2 determines if the temperature is above 102°C. This rung has another Compare instruction, but this one goes TRUE if word A is greater than word B. Notice that word A is again the actual temperature (from the ADC), and word B is the upper-limit temperature of 102°. Therefore, the rung will go TRUE if the oven temperature is greater than 102°. The value of this rung is stored in bit 0 : 3/1.
3. Rung 3 activates the heating element via the control logic that turns the heating element on and off—that is, the OUTPUT instruction *Heater* (0 : 3/2) directly controls the heating element. The rung will go TRUE if the on-off switch is on, *and* the temperature is not over 102° (notice this is an NC instruction), *and* the temperature is less than 98°. Once *Heater* is on, it is sealed on with the parallel branch 0 : 3/2 so that, even when the oven temperature rises above 98°, the rung will stay TRUE. With the heating element on, the oven temperature will eventually rise to above 102°, and so the NC instruction (*Temperature > 102°*) will go FALSE, breaking the seal and turning *Heater* off. The rung will stay FALSE until the temperature drops below 98°, and then the cycle starts over.

## Advanced Instructions

So far, we have just examined the basic instructions supported by virtually any PLC. PLCs are getting more sophisticated, and as they do, their instruction sets expand. In fact, many newer PLCs more closely resemble a real-time process-control computer than simply a substitute for relay logic. Many of the advanced instructions operate on a digital word instead of a single bit. Some of these words may be digitized analog data (such as temperature) used with an analog I/O module, which is an ADC or a DAC. Counter and timer accumulation values are stored as words. In many cases, the purpose of processing these number quantities is to arrive at a yes-no decision, where the actual output of the program is still in terms of individual bits (that is, turning a motor on or off). Then too, it is also possible to output a whole digital word to be converted to an analog voltage by an I/O module. Such a voltage could be used to control a motor's speed.

The higher level instructions come in various categories:

- **Comparison instructions** compare two words and yield a single bit as a result. For example, a word may be tested to see if it is larger than another word or to see if it is equal to another word. These instructions are demonstrated in Example 12.6.

- **Math instructions** perform mathematical operations on words in memory. These operations may include addition, subtraction, multiplication, division, and others. A math instruction might be used to multiply a sensor input word by a constant, for scaling purposes.
- **Logical and shift instructions** perform logical operations such as AND, OR, and NOT and Right and Left Shifts on words stored in memory. For example, the AND instruction could be used to mask out certain bits in a word.
- **Control instructions** allow for such things as jumps and subroutines. For example, a Jump instruction, when TRUE, will cause the execution to jump ahead (or back) to any designated rung.
- **PID control** is available on some PLCs. In some cases, it is built into an I/O module; in other cases, it is programmed as an instruction in the ladder diagram. The PID instruction is covered in the next section.
- **I/O message and communications instructions** allow for PLCs to send and receive messages and data to other PLCs or a terminal (PC), for those applications where PLCs are networked together.

### PID Instruction

PID control is becoming more commonly available on PLCs. In most cases, the PID function is implemented by an instruction and uses regular I/O (analog) channels. Another possibility is a separate PID hardware module, in which case it is actually an independently operating controller running under supervisory control of the PLC.

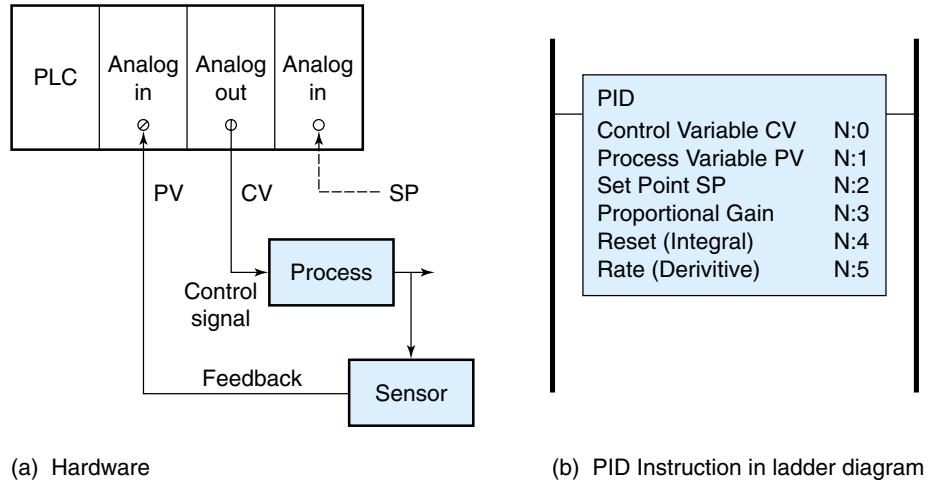
The PID instruction implements the PID control strategy discussed in Chapter 11. It would be used in a closed-loop system to control some physical parameter such as temperature, liquid level, or even motion (although a motion problem may require faster responses than the PLC can deliver).

A sample PID instruction is shown in Figure 12.26(b). It consists of a box or circle with the pertinent information placed in or around it. It is usually on a rung all by itself (there is no conditional logic on the left side of the rung), which means the PID instruction is always TRUE. Thus, as long as the PLC is running, the PID function will be trying to maintain some physical parameter at its set point. The hardware to support the PID instruction is shown in Figure 12.26(a) and consists of

- One I/O analog channel for the output signal  $CV$  (control variable)
- One input I/O channel (analog) for the feedback signal  $PV$  (process variable)
- Possibly an input I/O channel (analog) for the set point  $SP$  (However, the  $SP$  does not necessarily need a physical input, because it can be set by another instruction or is simply a constant.)

The instruction symbol specifies the addresses in RAM of the PID parameters. As shown in Figure 12.26(b), these memory locations store (at a minimum) the current values of  $CV$ ,  $PV$ , and  $SP$ , as well as the PID constants for proportional, integral and derivative ( $K_p$ ,  $K_I^*$ ,  $K_D^*$ ).

**Figure 12.26**  
PID instruction.



Tuning the PID controller means finding the values of  $K_p$ ,  $K_I$  and  $K_D$  that cause the system to remain stable, respond quickly to disturbances without excessive overshoot, and have a minimum of steady-state error. The method of tuning suggested by Allen-Bradley is similar to the continuous-cycle method discussed in Chapter 11 (see Example 11.6). This method is as follows:

1. Initially set  $K_p = 1$ ,  $K_I = 0$ ,  $K_D = 0$
2. Slowly increase the gain ( $K_p$ ) until the system goes into a steady oscillation, and record the period of oscillation (period =  $T_C$ )
3. Initial settings: set  $K_p$  to half of what was needed to go into oscillation, set  $K_I = 1/T_C$ , and set  $K_D = T_C/8$ .

## Other PLC Programming Languages

Ladder diagram programs are highly symbolic and are the result of years of evolution of industrial control circuit diagrams. The PLC is a relative latecomer, and it was adapted to use traditional ladder logic. However, as we have noted, a PLC program is not actually a wiring diagram but a way to describe the logical relationship between inputs and outputs. There are now more concise ways to convey control logic than ladder diagrams. These programs look more like conventional computer programs, using words and mathematical symbols instead of pictures.

One type of PLC language used by Allen-Bradley is called simply *ASCII programming*, or “creating an ASCII archive file.” Basically, this is a method of conveying a ladder diagram by using words instead of pictures. Each instruction is given a three-letter designation called a **mnemonic**. Table 12.3 lists the more common instructions. In Table 12.3, the first three mnemonics represent such devices as switches and relays, and the last three mnemonics specify how the devices are interconnected logically. In an ASCII

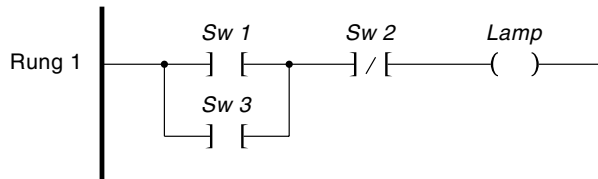
**TABLE 12.3**

**ASCII Programming Instructions**

Symbol	Mnemonic	Explanation
—] [—	XIC (examine if closed)	Goes TRUE when the contacts close.
—] / [—	XIO (examine if open)	Goes TRUE when the contacts open.
—( )—	OTE (output energize)	Energizes the relay when TRUE.
—┬—	BST (branch start)	Starts a parallel branch.
└┬┘	NXB (next branch)	Ends previous branch, and starts new branch.
——┘	BND (branch ends)	Ends the parallel branches.

**Figure 12.27**

A ladder diagram and the corresponding ASCII program.



**Program Instructions (ASCII)**

**Explanation**

!RUNG 1	Identifies the rung.
SOR BST XIC Sw 1 NXB	Start of the rung; branch starts and XIC Sw 1 is on the top branch, the next branch follows.
XIC Sw 3 BND	On the next branch is XIC Sw 3, and then this branching ends.
XIO Sw 2 OTE Lamp	On the main rung is XIO Sw 2, then a relay driving a lamp.
EOR	This rung is finished.

**Program Instructions (Boolean)**

**Explanation**

START Sw 1	Switch 1 is first logical variable.
OR Sw 3	Switch 3 is in parallel with Switch 1 (which is the same as an OR operation).
AND	The previous result will be ANDed with
NOT Sw 2	the inverse of the state of Switch 2.
OUT Lamp	The result of the logic equation is used to activate the Lamp output.

program, the ladder diagram is conveyed by placing the mnemonics and addresses in a certain order. These concepts are best shown by Figure 12.27, a simple ladder diagram followed by the corresponding ASCII program.

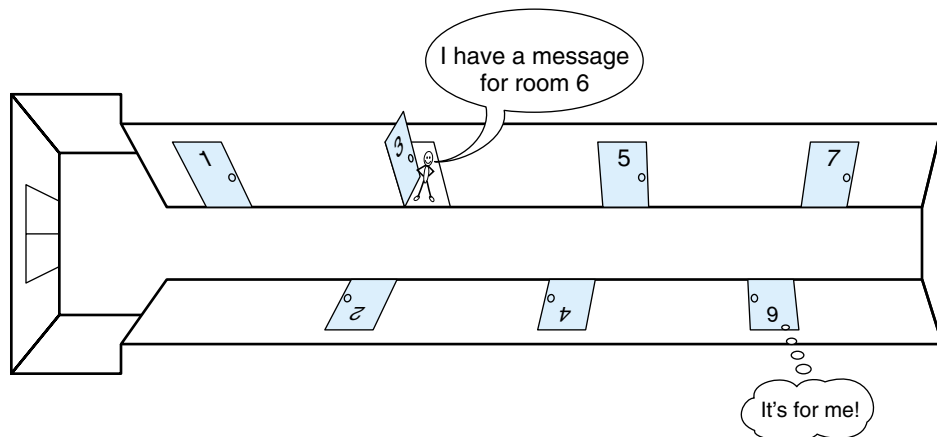
Another type of PLC programming language uses Boolean logic directly. This involves listing commands such as AND, OR, and NOT in such a way as to describe the program logic. An example of this approach is shown in Figure 11.27. Still others PLCs can be programmed with high level languages such as BASIC or C.

## 12.4 PLCS AND NETWORKS

Networks are increasingly being used to interconnect the components of control systems with each other and the rest of the factory. Physically, a network is a wire acting as an “electronic highway” that can pass messages between **nodes** (PCs and other electronic devices). Each node on the network has a unique address, and each message—called a **data packet**—includes the address of where it’s going and where it came from. All data on the network is sent serially (one bit at a time) on one wire. The most common type of network uses the **bus topology**, which means that all the nodes tap into a single cable (see Figure 12.31). The bus type of network is analogous to a long hall in a hotel, with room doors on either side, as illustrated in Figure 12.28. Each room has a room number (address), and a message could be passed from one room to another by shouting it down the hall. For example, if the person in room 3 wanted to send a message to the person in room 6, he or she would open the door and shout, “I have a message for room 6. The message is blah, blah, blah.” All the other occupants would be listening at their doors, but only the occupant of room 6 would actually take note and receive the message. Besides conveying the general concept, this analogy illustrates two other points about a bus type of

**Figure 12.28**

Analogy for a bus type of network: a long hall in a hotel, where each room is a node.

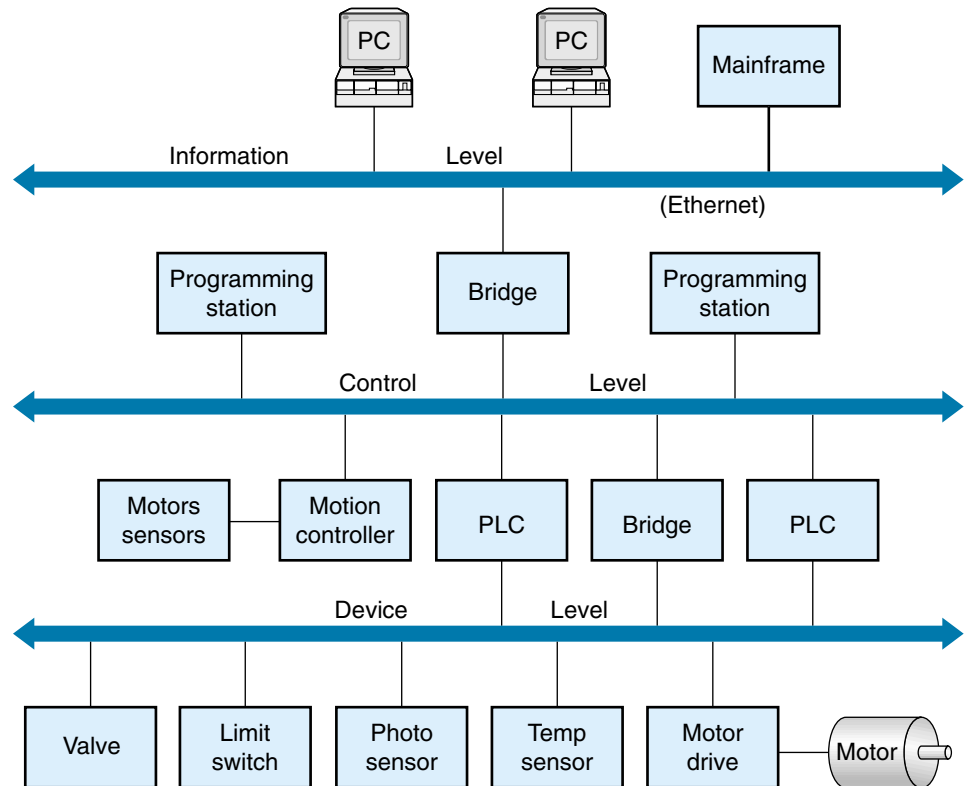


network: Every message is broadcast to every door, and only one message can be communicated at a time.\*

There are three levels of networks that could be used by industrial organizations (see Figure 12.29), and all three levels can be interconnected so that any node can communicate with any other node in the building. The highest level is Ethernet. **Ethernet** is the traditional network used for interconnecting PCs and mainframe computers for the purpose of exchanging data and data files; this is known as the *information level*. Ethernet is good for communicating financial, inventory, and production data, but it may have difficulty dealing with a real-time process like a PLC responding quickly to a change in motor speed. The problem is that Ethernet is not a deterministic network. A **deterministic network** is able to assign priority to a message and guarantee that it will arrive in a given time. Ethernet may slow down significantly in times of heavy traffic.

**Figure 12.29**

Three levels of networks.



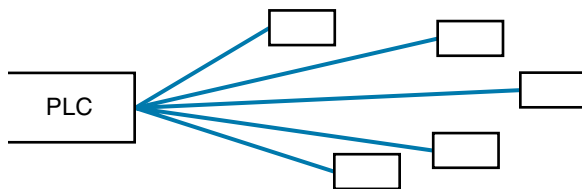
\*The integral constant  $K_I$  is usually called *Reset*  $= 1/T_C$ , and the derivative constant  $K_D$  is usually called *Rate*  $= T_D$

\*This is true for most networks that use a digital signal in *baseband* (no frequency multiplexing).

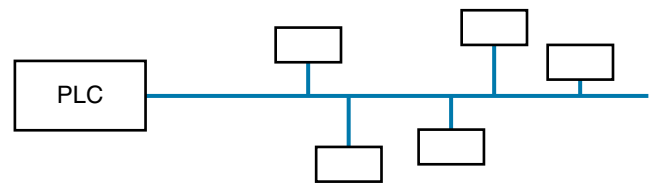
The second network level is known as the *control level*. A network at the control level provides deterministic communication and would be used to interconnect PLCs with other PLCs and with their supervisory computers and programming stations (typically PCs). A network at the control-level may use a *token system*\* to guarantee that each node that needs to transmit will get an opportunity to do so. An example of this type of network is Control Net (Allen-Bradley).

The third network level is known as the *device level*. A **device-level network** is used to connect low-level devices such as sensors and actuators to a PLC. Traditionally, individual sensors are connected directly to the PLC, and it might seem like “overkill” to use a network for this job, but there are three good reasons for using a network:

1. A device network simplifies wiring. Figure 12.30 compares the traditional point-to-point sensor wiring with the device network. Clearly the network is a simpler system that uses less wire. In fact, many new cars use a device network to connect electrical devices (locks, controls, and the like); this reduces the amount of wiring needed.
2. With a network, the sensor data arrives in better shape. In the traditional point-to-point system, a low-level analog voltage may have to travel many feet. The signal is subject to attenuation and noise. Even the current-loop system is subject to noise and other losses. With a device network, the analog voltage is converted to digital *at the source* and transmitted as a digital message with error detection. There is no chance that the signal will be attenuated (as long as the digital data gets through), and any corrupted data can be identified as such and re-transmitted.
3. Network devices tend to be more “intelligent.” Once a device has electronics built in to communicate on the network, it’s a small step to build in more functionality with little additional cost. For example, a photo cell could send a message saying the light level has diminished, (indicating that the lens may be getting dirty or that someone has bumped it out of position).



(a) PLC connected to sensors and actuators using point-to-point wiring



(b) PLC connected to sensors and actuators using a device-level network

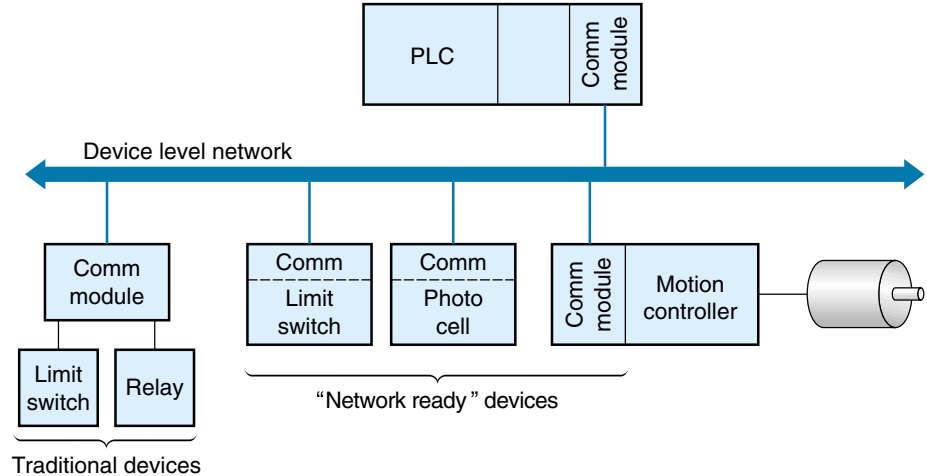
### Figure 12.30

Connecting a PLC to devices.

\*A token is a small data packet that is passed from node to node. There is only one token, and a node can transmit one chunk of data only if it has the token. This system guarantees that each node gets a turn to send data.

**Figure 12.31**

Each device connected to the network must use a network interface circuit (communication module).



A device-level network may be able to communicate in different ways, depending on the needs of the application. Probably the most common system is polled I/O. An example of **polled I/O** occurs when a PLC asks (polls) a sensor for data only when it needs it. In a **change-of-state** system, a sensor would initiate a data transfer only when there was a change to report. For example, a switch on a furnace door would only report when the door was actually opened. In a *cyclic* system, the sensor would send its data at a predefined rate, such as once every 100 ms. Finally, in a *strobe* system, one message is delivered to many devices simultaneously.

More equipment and procedural steps are required for setting up a device network and getting it operational, compared with the traditional wiring system of Figure 12.30(a). First, each device must connect to the network through an interface unit (see Figure 12.31). Traditional simple devices such as limit switches, photocells, and temperature sensors must connect to the network through a communication module. The communication module enables the operator to set the address of the device, perhaps with a thumbwheel switch. More sophisticated devices, such as motor drive controllers and other “network ready” devices, have the network communication circuitry built in. Also, to make the system work, the PLC program may have to include instructions to send initialization commands at start-up to those devices that need it, and it must be programmed to communicate with its devices through the network. Examples of device-level networks include **DeviceNet** (Allen-Bradley), and the **Fieldbus**.

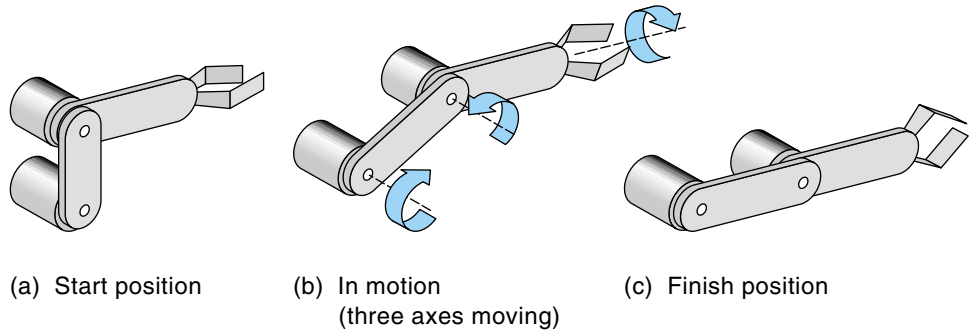
## 12.5 MOTION CONTROLLERS

**Motion controllers** are control circuits specifically made to coordinate movement. A typical application would be a robot picking up an object from one place, moving to another place, and putting the object down. As shown in Figure 12.32, this action could involve



**Figure 12.32**

In this example of motion control, three axes are moving at the same time.

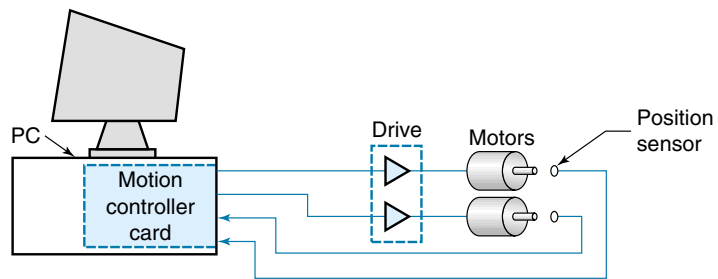


moving two or more axes at the same time. The motor for each axis would accelerate up to speed, travel at a fixed speed to the vicinity of the destination, and then decelerate to a stop. The speeds of both motors would be controlled so that they both arrived at the destination at the same time, even if that meant that they moved at different speeds.

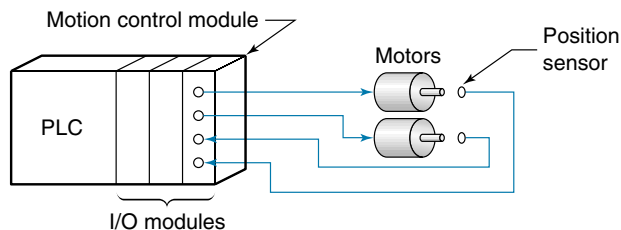
A motion controller is basically a programmable digital controller similar to those already discussed. A typical controller is built on a circuit card and may be able to control 4 to 8 axes. The controller can be either open-loop or closed-loop. Open-loop systems have to use stepper motors. Closed-loop systems use DC or AC servo motors and require feedback position sensors, which are typically optical encoders. Outputs from the motion controller card provide the speed or torque signal to the motors (or motor drives); inputs to the controller come from the sensors. Two possible hardware configurations are shown in Figure 12.33. The system shown in Figure 12.33(a) is a PC with

**Figure 12.33**

Motion controller hardware.



(a) Motion controller on a PC expansion card



(b) Motion controller is a PLC module

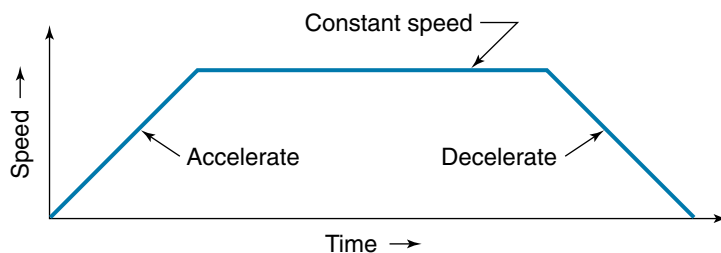
a motion controller expansion card. The card can be programmed with special software that runs under Windows. In another configuration, shown in Figure 12.33(b), the motion controller is a module that runs under a PLC. In this case the PLC would download the motion parameters into the motion controller, and the motion controller would then work on its own.

Like virtually all digital controllers, a motion controller is basically a computer executing a program. What makes it a “motion controller” is that the hardware and software have been adapted specifically to facilitate controlling *motion*. A processor in a motion controller could be either a general-purpose microcontroller or a digital-signal processor (DSP). A **DSP** is a processor that is designed to support high-performance, repetitive, numerically intensive tasks. For example, DSPs may be able to do two multiply operations at the same time, can perform tight, quick loops, and have irregular instructions sets that allow several instructions to be encoded in the same instruction. In other words, a DSP is good for control operations where multiple-axis operation and quick response are required. However, regardless of the type of processor used by the motion controller, the basic setup and operation are the same:

1. By means of a PC and proprietary software supplied or specified by the manufacturer of the motion controller, a series of commands are entered that describe the desired motion.
2. The motion commands are processed by the software into “machine language” and downloaded into memory on the motion processor card.
3. When activated, the processor on the motion controller card executes the machine-language program loop. It is this program that takes over the job of actually processing the feedback signals and sending motor-control signals to the motor.

The motion controller may apply several control strategies in a sequence when moving to its target. A typical move would start by accelerating to a designated speed, then moving toward the target at that speed, and finally decelerating and possibly using PID to close in on the final position. This particular sequence is sometimes called *trapezoid* motion because a graph of the motion resembles a trapezoid (see Figure 12.34). Other ways of getting to the target (depending on the system) might be called *rapid*, *circular*, or *S-curve*. The programmer can specify whether more than one axis is to move at the same time, and if so, the controller will move them in a coordinated fash-

**Figure 12.34**  
Graph of “trapezoid”  
motion



ion, all finishing at the same time. If more than one move is specified in succession with no pause in between, the first move will blend smoothly into the second, and so on, until the target is reached.

There is not yet a standard accepted programming language for motion controllers, which means that most systems have their own unique language. An example of a simplified program for the Delta-Tau PMAC system follows.

<b>Command</b>	<b>Explanation</b>
#1->1000X	Assign motor #1 to X-axis.
#2->1000Y	Assign motor #2 to Y-axis.
CLEAR	Erase existing contents.
LINEAR	Specify motion mode (trapezoid).
TA500	Set acceleration value.
TS200	Set speed.
X0 Y0	Move to start position.
X10 Y0	Move to $x = 10$ and $y = 0$ .
DWELL	Wait 500 ms.
X0 Y10	Move to $x = 0$ and $y = 10$ .
X0 Y0	Move back to start position.

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## SUMMARY

One common type of control system governs the repetitious sequence of events. Traditionally, this kind of control was implemented with electromechanical relays, timers, and sequencers. Relays can be interconnected to provide some control logic—for example, two relays in series become an AND function. Time-delay relays will delay activating their contacts for some period of time (after being energized). Sequencers use a rotating cam to open and close contacts in some sequence.

A ladder diagram, a special kind of wiring diagram, was developed to document electromechanical control circuits. This type of diagram (which resembles a ladder) has two vertical wires (rails) on either side of the drawing; these wires supply the power. Each rung of the ladder diagram connects from one rail to the other and is a separate circuit, which typically consists of some combination of switches, relay contacts, relay coils, and motors. It is common for the coil of a relay to be in one rung and the contacts to be in another.

In the late 1960s, the programmable logic controller (PLC) was developed to replace electromechanical controllers. The PLC is a small, microprocessor-based, process-control computer that can be connected directly to switches, relays, small motors, and so on, and is built to withstand the industrial environment. The PLC executes a program that must be written for each specific application. The program is always in the form of a loop, which has the following pattern: The PLC reads and stores

the input switch and sensor data; it determines what the output(s) should be; it sends the output control signals to the devices to be controlled.

To use a PLC requires setting up the hardware and software. The hardware installation consists of wiring the PLC to all switches and sensors of the system and to such output devices as relay coils, indicator lamps, or small motors. The control program is usually developed on a PC, using software provided by the PLC manufacturer. This software allows the user to develop the control program in the form of a ladder diagram on the monitor screen. Once the program is complete, it is automatically converted into instructions for the PLC processor. The completed program is then downloaded into the PLC. Once the program is in the PLC's memory, the programming terminal can be disconnected, and the PLC will continue to function on its own.

Increasingly PLCs are being interconnected with networks. There are three levels of networks: the *information level* for office communications, the *control level* for interconnecting PLCs, and the *device level* for connecting sensors and actuators to the PLC.

Motion controllers are control circuits specifically made to coordinate movement. A typical motion controller works with a closed-loop system to coordinate the movement of two or more axes simultaneously.

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## GLOSSARY

**analog I/O module** A module that plugs into the PLC and converts real-world analog signals to digital signals for the PLC and vice versa.

**bit instruction** The basic PLC instruction that logically represents a simple switch or relay coil; the state of each switch or coil occupies 1 bit in memory.

**bus topology** A type of network wherein all the nodes tap into a single cable at different places, and the cable has a beginning and an end (that is, the cable is not in a ring).

**change-of-state system** A method of communicating on a network wherein a device sends a message only when something changes.

**comparison instruction** Compares two words and yields a single bit as a result.

**control instruction** Allows for such things as jumps and subroutines.

**data file** The collection of data in a PLC that represents the present condition of switches, counters, relays, and the like.

**data memory** The section of RAM memory in a PLC where the data file is stored.

**data packet** A message transmitted on a network; usually includes the address of where it's going, the address of where it came from, and the message itself.

**deterministic network** A network able to assign priority to a message and to guarantee that it will arrive in a given time.

**device-level network** A network used to connect low-level devices such as sensors and actuators to a PLC.

**DeviceNet** A type of device-level network developed by Allen-Bradley. See **device-level network**.

**Digital-signal processor (DSP)** A type of digital processor that is designed to support high-performance, repetitive, numerically intensive tasks.

**discrete I/O** I/O that deals with on-off devices. Discrete input modules are used to connect real-world switches to the PLC, and discrete output modules are used to turn on lamps, relays, motors, and so on.

**electromechanical counter** A device that counts events. Each time it receives a pulse, it increments a mechanical counter, which can then be read by an operator or used to activate some other process.

**electromechanical relay (EMR)** A device that uses an electromagnet to close (or open) switch contacts—in other words, an electrically powered switch.

**electromechanical sequencer** A device that provides sequential activation signals; it works by rotating a drum with cams, which activates switches.

**EMR** See **electromechanical relay**.

**Ethernet** The type of network used to interconnect PCs and mainframe computers for the purpose of exchanging data and data files.

**Fieldbus** A type of device-level network. See **device-level network**

**I/O message and communications instructions** Allow PLCs to send and receive messages and data to other PLCs or a terminal (PC), for those applications where PLCs are networked.

**Input/Output modules** The connection points where real-world switches, relays, and the like are connected to the PLC.

**ladder diagram** A type of wiring diagram (which resembles a ladder) used for control circuits; ladder diagrams typically include switches, relays, and the devices they control.

**latching** See **sealing**.

**logical and shift instructions** Perform logical operations such as AND, OR, and NOT and Right and Left Shifts on words stored in memory.

**math instructions** PLC mathematical operations on words in memory; these operations may include addition, subtraction, multiplication, division, and others.

**mnemonic** An English-sounding abbreviation for PLC logical instructions.

**Motion controller** A control circuit specifically made to coordinate movement. A typical motion controller works with a closed loop system to coordinate the movement of two or more axes simultaneously.

**node** Any unit connected to a network that can transmit and receive data.

**normally closed (NC) contacts** Relay contacts that are closed in the de-energized state and open when the relay is energized.

**normally open (NO) contacts** Relay contacts that are open in the de-energized state and closed when the relay is energized.

**off-delay relay** A time-delay relay that, after being de-energized, waits a specified period of time before activating.

**on-delay relay** A time-delay relay that, after being energized, waits a specified period of time before activating.

**PID control** A common control strategy. PID instructions are available on some PLCs.

**PLC** *See* **programmable logic controller**.

**PLC bus** The wires in the backplane of a modular PLC system that allows the microprocessor (within the PLC) to communicate with plugged-in modules.

**plug-in module** An interface circuit to the PLC, packaged as a separate module, which plugs into the PLC. Different modules support different I/O devices, from simple switch and relay output modules to ADC and DAC modules.

**pneumatic time-delay relay** A time-delay relay that generates the delay by allowing air to bleed out of a spring-loaded bellows.

**polled I/O** Probably the most common method of communicating on a network, wherein a PLC asks (polls) a sensor for data only when it needs it.

**power rails** Part of a ladder diagram, the two vertical lines on each side of the diagram, which represent power lines.

**processor** Part of the PLC that executes the instructions; it is a small, microprocessor-based computer.

**program file** Contains the program that the PLC executes. The program file is part of the PLC memory (the other part being data files).

**programmable logic controller (PLC)** A small, self-contained, rugged computer, designed to control processes and events in an industrial environment—to take over the job previously done with relay logic controllers.

**program memory** When applied to PLCs, the section of memory (RAM or EEPROM) where the PLC program is stored.

**programming port** A connector on a PLC through which the program is downloaded.

**run mode** The operational mode of the PLC when it is actually executing the control program and hence performing some control operation.

**scan** The event when the processor makes one pass through all instructions in the control program loop.

**sealing** A relay wired such that once energized its own contacts take over the job of providing power to the coil; it will stay energized, even if the original energizing signal goes away.

**solid-state time-delay relay** A time-delay relay that generates the delay with an electronic circuit.

**thermal time-delay relay** A time-delay relay that generates the delay by allowing a bimetallic strip to heat and bend, which activates a switch.

**time-delay relay** A relay with an intentional delay action; that is, the contacts close (or open) in a specified period of time *after* the relay is energized (or deenergized).

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## EXERCISES

### Section 12.1

1. Draw a relay wiring diagram for a circuit that implements the simple logic diagram shown in Figure 12.35. The circuit should be in the style of Figure 12.1(a).
2. Repeat Exercise 1 except draw the circuit as a ladder diagram.
3. A small house has three windows and two doors. Each window and door has a switch attached such that the contacts close when a door or window opens. Draw a ladder logic diagram that will turn on a light if one or more windows are open or if both doors are open.
4. In a processing plant, jars on a conveyer belt are cleaned out with a high-pressure air jet just prior to being filled (Figure 12.36). When a jar approaches the cleaning station, it activates a switch (with both NO and NC contacts). The conveyer belt stops for 10 s while the air jet is on; then the conveyer belt starts, and the jar moves along. Draw a ladder logic diagram to control this process.

Figure 12.35

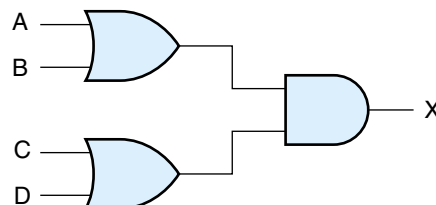
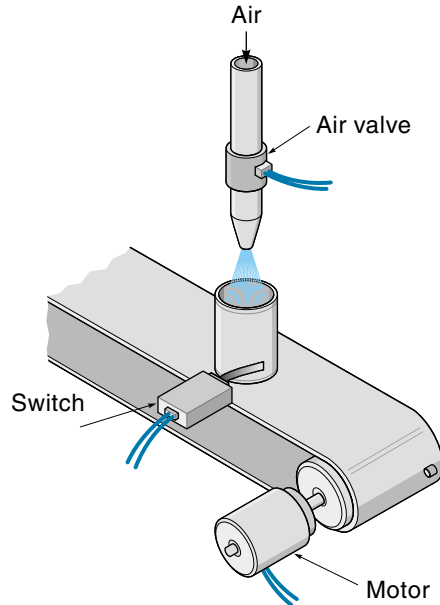


Figure 12.36



### Section 12.2

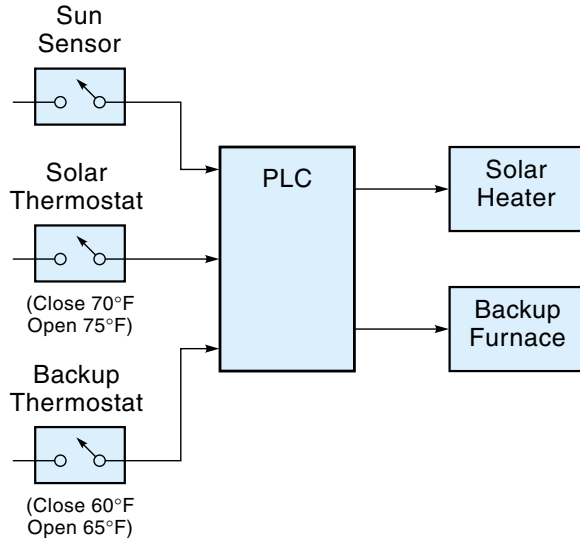
5. You just bought a PLC, and it's still in the box. List the general steps you will have to take to get the PLC operational in a specific task.
6. A PLC has eight inputs and eight outputs as shown in Figure 12.17(a). Draw a wiring diagram for this PLC that will be used as the controller for the situation of Exercise 3.
7. A PLC has eight inputs and eight outputs as shown in Figure 12.17(a). Draw a wiring diagram for this PLC that will be used as the controller for the situation of Exercise 4.
8. List the steps the PLC takes to execute the ladder diagram program.

### Section 12.3

9. Draw a PLC ladder diagram for the situation described in Exercise 3.
10. A motor is controlled by a NO start button, a NC stop button, and an overload device. The overload device is normally closed and opens if the motor overheats. Draw the PLC ladder diagram for the circuit. Include a light that comes on when the motor overheats.
11. A PLC is to control the solar heating system shown in Figure 12.37. The system has two interrelated parts: (a) A solar thermostat turns on and off the solar heater if the sun sensor says the sun is shining, and (b) a backup thermostat turns on and off a conventional furnace if the solar energy is insufficient. Both heating systems share the same ductwork, so if the backup thermostat closes, the PLC *must* turn on



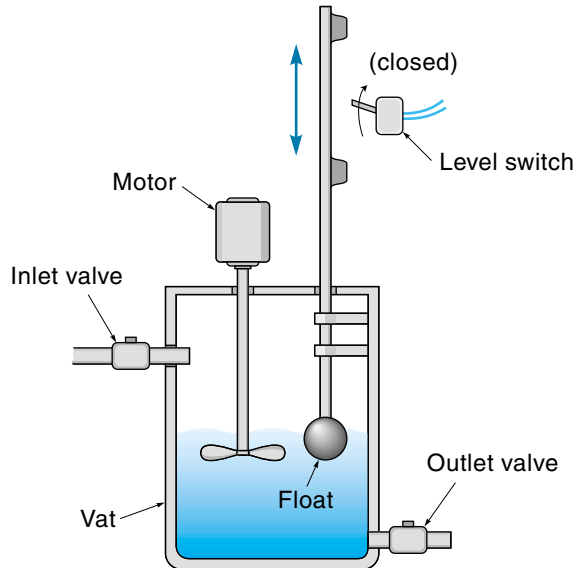
Figure 12.37



the backup furnace (and turn off the solar heater if it's on). Draw a PLC ladder diagram for this system.

12. A mixing vat has an inlet valve, an outlet valve, a mixing motor, and a single level-detecting switch (Figure 12.38). Both valves are opened by solenoids. The level switch closes when the vat is full and stays closed until the vat is empty. Draw a PLC ladder logic diagram to do the following:

Figure 12.38



- a. When the start button is pushed, the inlet valve opens until the vat is full.
  - b. The mixer motor comes on for 5 min.
  - c. The outlet valve opens until the vat is empty.
13. Modify the two-point control program of Example 12.6 so that the oven temperature limits are  $72\text{--}76^{\circ}\text{C}$ . The system should be started and stopped with NO push-button switches (instead of toggle switches).

### Section 12.4

14. What are the three levels of networks that might be found in a process-control industry.
15. What is a “device-level” network, and what advantages might this system offer over traditional point-to-point wiring?

### Section 12.5

16. How is a “motion controller” different from a PLC?
17. Using the sample program in Section 12.5 as a guide, write a motion-control program for a three-axis system to do the following:  
Start at  $X = 0$ ,  $Y = 0$ , and  $Z = 0$ .  
Move to  $X = 5$ ,  $Y = 5$ , and  $Z = 2$ .  
Move to  $X = 4$ ,  $Y = 4$ , and  $Z = 3$ .  
Pause 500 ms.  
Move to start position.

# APPENDIX **A**

## Thermocouple Tables for Type J (°F and °C)

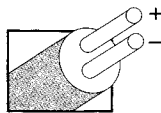
“© Copyright 1999 Omega Engineering, Inc. All rights reserved. Reproduced with permission of Omega Engineering, Inc., Stamford, CT 06907.”



Revised Thermocouple Reference Tables

TYPE J

Reference Tables N.I.S.T. Monograph 175 Revised to ITS-90



Thermocouple Grade

MAXIMUM TEMPERATURE RANGE

Thermocouple Grade 32 to 1382°F 0 to 750°C

Extension Grade

32 to 392°F 0 to 200°C

LIMITS OF ERROR

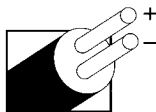
(whichever is greater) Standard: 2.2°C or 0.75% Special: 1.1°C or 0.4%

COMMENTS, BARE WIRE ENVIRONMENT: Reducing, Vacuum, Inert; Limited Use in Oxidizing at High Temperatures; Not Recommended for Low Temperatures

TEMPERATURE IN DEGREES °F REFERENCE JUNCTION AT 32°F

Iron vs. Copper-Nickel

Extension Grade



Thermoelectric Voltage in Millivolts

Table of thermoelectric voltages for Type J thermocouples, with °F on the x-axis (0 to 510) and mV on the y-axis. The table contains numerical values for each degree increment.

Table of thermoelectric voltages for Type J thermocouples, with °F on the x-axis (510 to 1600) and mV on the y-axis. The table contains numerical values for each degree increment.

°F 0 1 2 3 4 5 6 7 8 9 10 °F

°F 0 1 2 3 4 5 www.4electron.com °F



# APPENDIX **B**

*The Getting Started Guide for APS, Chapter 2 Control Basics*  
(tutorial on using the file system, addressing, and ladder logic for the  
Allen-Bradley SLC 500 and MicroLogix 1000 PLCs) 568

*Getting Results Guide for RS Logix 500, Chapters 1 and 5*  
(Instructions for using Windows based ladder logic programming for  
Allen-Bradley SLC 500 and MicroLogix 100 PLCs) 574

# Getting Started Guide

## Chapter 2

## Control Basics

This chapter introduces you to basic concepts essential for understanding how the SLC 500 controller operates. It covers:

- SLC 500 file concepts
- How external I/O devices communicate with the processor
- Addressing external I/O
- Ladder logic concepts

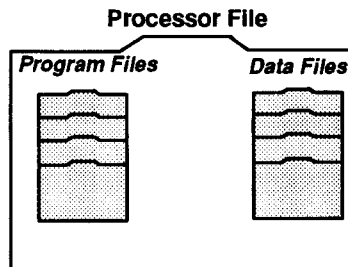
### SLC 500 File Concepts

The CPU, or processor, provides control through the use of a program you create. The program you create is called a processor file. This file contains other files that break your program down into more manageable sections. These sections are:

- Program Files – provide storage and control of the main program and subroutines.
- Data Files – contains the status of inputs, outputs, the processor, timers, counters, and so on.

### Processor Files

Each CPU can hold 1 processor file at a time. The processor file is made up of program files (up to 256 per controller) and data files (up to 256 per controller).



Processor files are created in the offline mode using APS. These files are then restored, also referred to as downloaded, to the processor for online operation.



## Program Files

Program files contain controller information, the main control program, and any subroutine programs. The first three program files are required for each processor file. These are:

- **File 0 – System Program**  
This file stores the controller configuration and other system information.
- **File 1**  
This file is reserved for internal controller use.
- **File 2 – Main Ladder Program**  
This file stores the main control program.
- **Files 3 – 255 – Subroutine Ladder Program**  
These files are optional and used for subroutine programs.

Most of your work with program files will be in file 2, the main program file. This file contains your ladder logic program which you create to control your application.

## Data Files

Data files contain the data associated with the program files. Each processor file can contain up to 256 data files. These files are organized by the type of data they contain. Each piece of data in each of these files has an address associated with it that identifies it for use in the program file. For example, an input point has an address that represents its location in the input data file. Likewise, a timer in the timer data file has an address associated with it that allows you to represent it in the program file.

The first 9 data files (0 – 8) have default types. You designate the remainder of the files (9 – 255). The default types are:

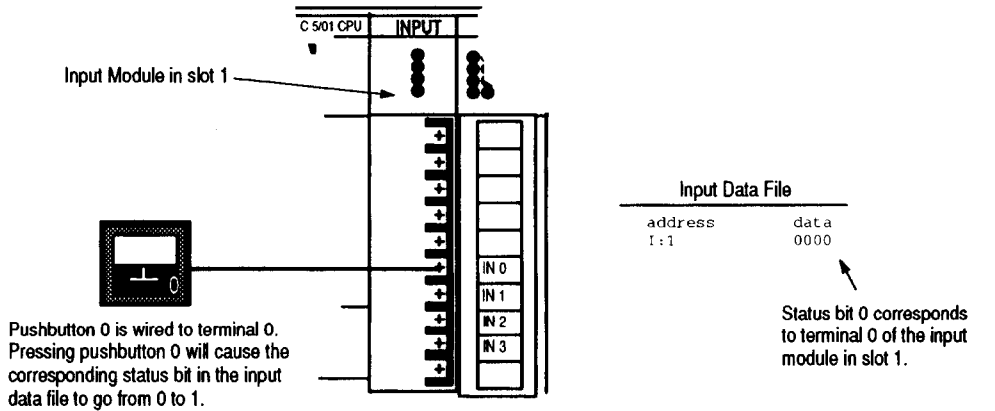
- **File 0 – Output Data**  
This file stores the state of the output terminals for the controller.
- **File 1 – Input Data**  
This file stores the status of the input terminals for the controller.
- **File 2 – Status Data**  
This file stores controller operation information.
- **Files 3 – 7**  
These files are pre-defined as Bit, Timers, Counters, Control, and Integer data storage, respectively.
- **File 8 – Float Data**  
This file is used by SLC 5/03™ with OS301 and SLC 5/04™ with OS400 processors for Float data storage.
- **Files 9 – 255**  
These files are user-defined as Bit, Timer, Counters, Control, Integer, and Float data storage.

Most of your work with data files will be in files 0 and 1, the output and input files. Refer to appendix A for an example of the Timer data file.

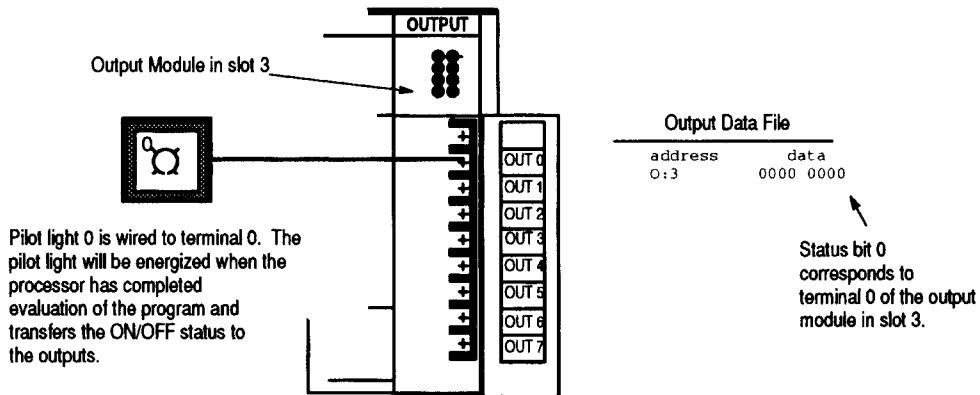
## How External I/O Devices Communicate with the Processor

The figure below applies to a modular controller demo unit having an input module in slot 1 and an output module in slot 3. See page 1–2 for a diagram of the slot location. To simplify the illustration, only pushbutton 0 and pilot light 0 of the external I/O are shown.

Each of the external input circuits is represented by a status bit in the input data file of the processor file. Each of the external output circuits is represented by a status bit in the output data file of the processor file. During controller operation, the processor applies the input data to the program, solves the program based on the instruction you enter, and energizes and de-energizes external outputs.



Closing an external input circuit changes the corresponding status bit from 0 to 1.  
 Opening an external input circuit changes the corresponding status bit from 1 to 0.



When an output data file status bit has been solved as a 1, the corresponding external output circuit will be energized (ON).  
 When an output data file status bit has been solved as a 0, the corresponding external output circuit is de-energized (OFF).

## Addressing External I/O

As pointed out in the last section, external inputs and outputs are linked to the input data file and output data file of the processor file. Each status bit in these files has an address. You specify the appropriate address when you enter an instruction in your ladder program.

For our purposes, input addresses have the form **I:e/b** where

**I** = Input data file  
**:** = Element or slot delimiter  
**e** = Slot number of the input module  
**/** = Bit or terminal delimiter  
**b** = Terminal number used with input device

Similarly, output addresses have the form **O:e/b** where

**O** = Output data file  
**:** = Element or slot delimiter  
**e** = Slot number of the output module  
**/** = Bit or terminal delimiter  
**b** = Terminal number used with output device

Examples:

**I:1/0** = Input, slot 1, terminal 0

**I:2/0** = Input, slot 2, terminal 0

**O:3/0** = Output, slot 3, terminal 0

**O:3/7** = Output, slot 3, terminal 7

**O:0/7** = Output, slot 0, terminal 7 (fixed controllers only because of slot 0)

**I:0/4** = Input, slot 0, terminal 4 (fixed controllers only because of slot 0)

Eventually, you will be addressing other data files, such as Status, Bit, Timer, Counter, Control, Integer, String, ASCII, and Float. Addressing of these files is discussed in the APS programming manual.

## APS Display of Instructions/Addresses

APS displays I/O addresses as shown below.

When you enter an XIC instruction (defined later) and the address I:1/0, APS will display the address with the instruction as follows:

```

      I:1.0
      ───┘ ┌──
           0
  
```

Explanation:

```

      I:1.0
      ───┘ ┌──
           0
  
```

Input data file, slot 1, word 0

Terminal 0

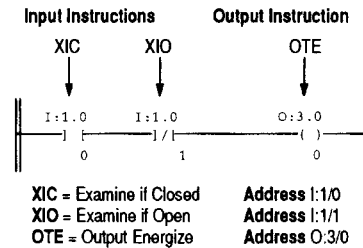
XIC instruction

## Ladder Logic Concepts

As we mentioned earlier, the program files you create contain the program used for your controlling application. The programs are written in a programming language called Ladder Logic. This name is derived from its ladder-like appearance.

A ladder logic program consists of a number of rungs, on which you place instructions. Instructions each have a data address associated with them and based on the status of these instructions the rung is solved.

The figure below shows a simple 1-rung ladder program. The rung includes two input instructions and an output instruction. Note, in the example below each instruction has a name (Examine if Closed), a mnemonic (XIC), and an address (I:1/0).



A simple rung, using bit instructions.

### True/False Status

The data file bits that these instructions are addressed to will be either a logic 0 (OFF) or a logic 1 (ON). This determines whether the instruction is regarded as “true” or “false”:

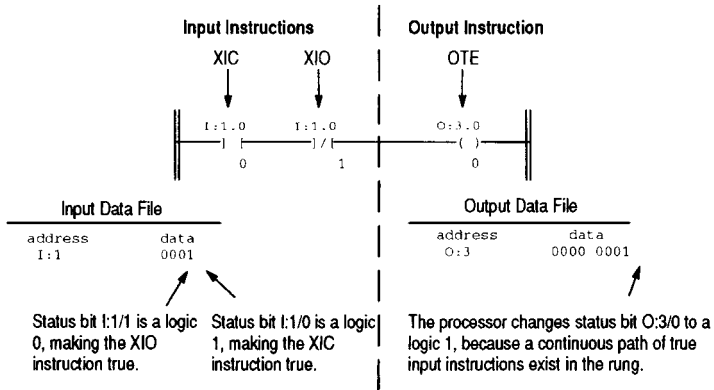
If the data file bit is	The status of the instruction is		
	XIC Examine if Closed —   —	XIO Examine if Open —  / —	OTE Output Energize —( )—
Logic 0	False	True	False
Logic 1	True	False	True

### Logical Continuity

During controller operation, the processor evaluates each rung, changing the status of instructions according to the logical continuity of rungs. More specifically, input instructions set up the conditions under which the processor will make an output instruction true or false. These conditions are:

- When the processor finds a continuous path of true input instructions in a rung, the OTE output instruction will become (or remain) true. We then say that “rung conditions are true”.
- When the processor does *not* find a continuous path of true input instructions in a rung, the OTE output instruction will become (or remain) false. We then say that “rung conditions are false”.

The figure below indicates the data file conditions under which the rung is true:



In the above example, if the input data file was 0000, then the rung would be false and the output data file would read as 0000 0000.

### The Processor Operating Cycle

The diagram below indicates the events that occur during the processor operating cycle. This sequence is repeated many times each second.

Event	Description	
Input Scan		The status of external input circuits is read. The input data file is updated with this information.
Program Scan		The ladder program is executed. The input data file is evaluated, the ladder rung is solved, and the output data file is updated.
Output Scan		The output data file information is transferred to the external output circuit, thus energizing or de-energizing it.
Communications		
Housekeeping	Processor internal housekeeping takes place.	

# Getting Started with RS Logix 500

## Welcome to RSLogix 500

The **RSLogix 500** software is a 32-bit Windows® ladder logic programming package for the SLC 500 and MicroLogix processors. Operating in the Microsoft® Windows 95 and Windows NT™ environment, **RSLogix 500** is compatible with programs created with any of Rockwell Software's DOS-based programming packages.

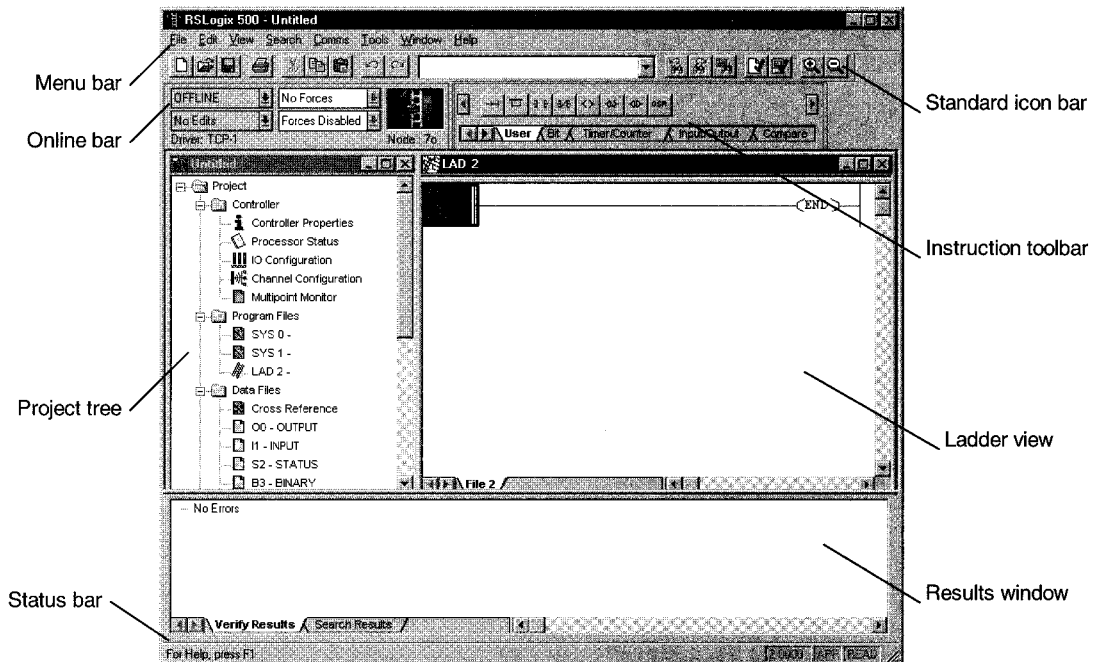
**RSLogix 500** software functionality includes:

- a free-form ladder editor that lets you concentrate on the application logic instead of proper syntax as you write your program
- a powerful project verifier that you use to build a list of errors you can navigate to make corrections at your convenience
- drag-and-drop editing to quickly move data table elements from one data file to another, rungs from one subroutine or project to another, or instructions from rung to rung within a project
- search and replace to quickly change occurrences of a particular address or symbol
- a custom data monitor to view separate data elements together and observe interactions
- a point-and-click interface called a project tree that lets you access all the folders and files contained in your project
- Trending and histogram functionality for monitoring and displaying process data
- SLC libraries for storing and retrieving portions of ladder logic for use across any of Rockwell Software's SLC programming software products.

## Exploring RSLogix 500

To navigate through the various windows and toolbars in **RSLogix 500** more easily, you should understand what they contain and what functionality each provides.

When you open a project in **RSLogix 500** you can expect to see:



- a project tree. Contains all the folders and files contained in your project. You can usually click an icon in this tree and then click the right mouse button for a menu that applies only to the icon selected. For example, if you click the right mouse button on a program file, you see options to rename the program file, open the program file, hide the program file, or reveal properties of the program file.
- the ladder view. In this part of the application window you can view several program files at the same time. This is where you edit your ladder logic.
- the results window. Displays the results of a Find All search or a verification procedure. You can hide this window view or separate it from the application window so that it can be placed anywhere on your screen.

- the menu bar. Select functionality from the menus that appear as you click each selection on this bar.
- the online bar. Learn the operational mode and see whether you have online edits or forces installed at a glance. You can even view the driver and node number.
- the standard icon bar. The standard icon bar contains many functions that you will use repeatedly as you develop and test your logic program. If you want to know what any of the icons represent, **RSLogix 500** can tell you. Just move your cursor to the icon. A floating ToolTip window will appear and tell you what the icon is used for.
- the tabbed instruction toolbar. Displays instruction mnemonics in tabbed categories. When you click on a category tab the instruction toolbar just above it changes to show that category of instructions. Click an instruction to insert it in your ladder program.
- the status bar. Look here for ongoing status information or prompts, as you use the software.

## System requirements

To use **RSLogix 500** software effectively, your personal computer must meet the following hardware and software requirements.

### Hardware requirements

The personal computer must have at least:

- an Intel Pentium™, Pentium-compatible, or 486 microprocessor
- 16 MB RAM for Fixed SLC 500, SLC 5/01, SLC 5/02 or MicroLogix 1000  
32 MB RAM for 8, 12, 16, 24, or 32Kbyte SLC 5/03, 5/04, or 5/05  
64 MB RAM for 64Kbyte SLC 5/03, 5/04, or 5/05
- 10 MB of free Hard Disk Space (or more based on application requirements)
- 16-color VGA Graphics Adapter 640 x 480 or greater resolution (256-color 800 x 600 optimal)
- a 3.5-inch, 1.4 MB disk drive
- any Windows-compatible pointing device

### Software requirements

The operating system must be Microsoft Windows 95 or Microsoft Windows NT v.4.0. Both operating systems are compatible with RSLinx/RSLinx Lite communication software.



- **RSLogix 500** relies on Rockwell **RSLinx™** ver 1.70.62 or later. One copy of the **RSLinx Lite** software is included with the **RSLogix 500** software.  
(**Note:** If you are currently using WINtelligent Linc, we recommend that you update to RSLinx unless either of the following conditions are true of your system configuration. Stay with WINtelligent Linc if you are using a Modem or if you have a channel configured for DF1-half duplex slave protocol.)

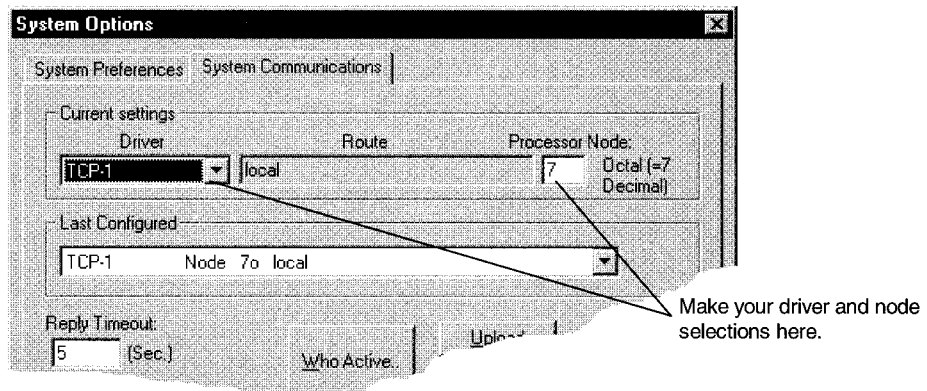
## Quick Start steps

The following steps explain how to get up and running quickly with **RSLogix 500** as soon as you install it.

### Step 1 ■ Configure system communications

Do this before you begin a new project. The settings you establish with this step will remain with the project and will be applied when you attempt to download any logic program.

From the Comms menu, click System Comms. Then click the System Communications tab on the System Options window, to give it focus.



From this tab select your driver, node, and timeout. A shortcut method is available to you if you have previously used the **RSLogix 500** software to set up communications with a network node. Click the Last Configured drop-down listbox to see a list of previously configured drivers and nodes. Then select from the list.

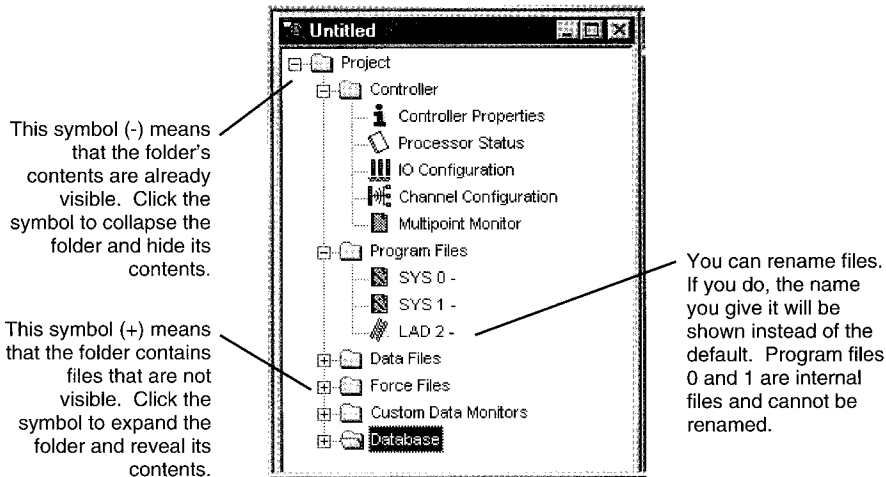
You can also access the Who Active feature available in **RSLinx**. This utility displays all available stations/nodes on your communication networks. This can help if you are unsure of the node number of a particular device on the network.

For more detailed information about the steps you follow to configure your system communications refer to chapter 3 in this book.

## Step 2 ■ Create a new project or open an existing project

### New Projects

**RSLogix 500** is based on projects. Projects are the complete set of files associated with your program logic. You create a project from the File menu by clicking New. **RSLogix 500** prompts you for the type of processor you will communicate with and creates a project tree control. This project tree is your access point to program, data table, and database files.



### Existing Projects

From the File Menu, click Open. Use the dialog that appears next, to Open a ladder logic project and/or its associated database. Depending on the type of action (open or import) **RSLogix 500** presents a default file extension. You may however, select a different file type to open or import. Make sure you have the correct type of file listed in the Files of Type drop-down listbox on the Open/Import SLC 500 Program window.

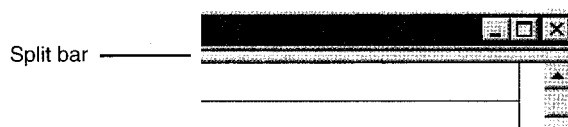
#### TIP



When you have the Open file window opened, press F1 for help that can assist you with clarification of the different file types and their extensions.

### Opening Multiple Files

To open multiple files within the same project, you can split the viewing window.

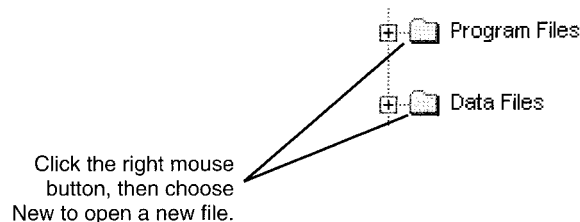


Use the mouse to point to the split bar. The pointer turns into a double bar with two arrows. Drag the bar up or down to its new position, allowing you to see two views of the window.

You cannot view program files from different projects with only a single **RSLogix 500** application running on your computer. You must open more than one application to work on multiple projects at the same time. Once you have the project opened, however, you can drag and drop instructions and data between them.

### Step 3 ■ Create program and data table files

The project tree is your point-of-entry for creating new files or accessing existing files. To create a new file, right-click the program or data file icon and then select **New** from the menu. You will be prompted for information about the file.



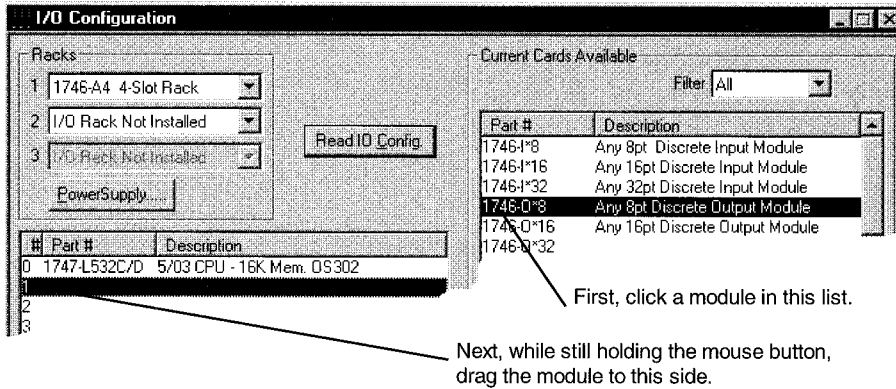
Program files contain controller information, the main ladder program, and any subroutine programs. With SLC controllers you may have up to 256 program files. With the MicroLogix controller you may have up to 16 program files.

Data table files contain the status information associated with external I/O and all other instructions you use in your main and subroutine ladder program files. In addition, these files store information concerning controller operation. You can also use the files to store recipes and look-up tables if needed.

### Step 4 ■ Define chassis and modules

After you have opened a project, you have to define your chassis, identify the I/O cards that you are using by indicating their slot position within the processor rack, and select a power supply for each rack in your configuration. A real application might have up to three racks and many I/O modules.

You perform these procedures in the I/O Configuration window. Access this window by double-clicking the I/O Configuration icon in the project tree. Then click a module in the list on the right side of the window and drag it into the slot where you want it to reside.

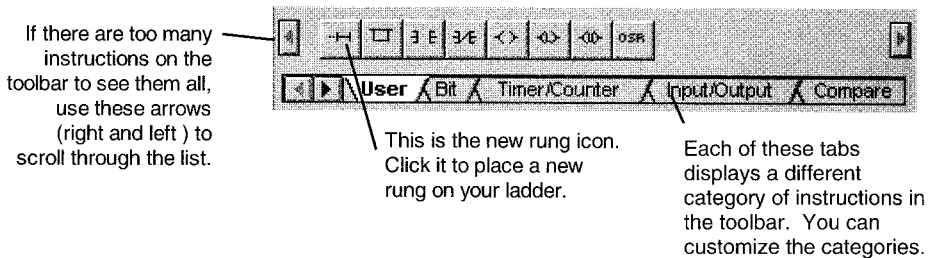


From the I/O Configuration window, click the Power Supply button to examine the loading on a rack based on the module configuration that you have selected.

### Step 5 ■ Enter a logic program

When you open a program file by double-clicking its icon in the project tree, the ladder file opens in the right half of the RSLogix 500 window. Usually program file #2, the main program file, will be opened when you open a project. If you have not begun to enter any ladder logic, only the end rung will show.

Click on the end rung and then select the new rung icon from the user toolbar. To place an instruction on a rung, click its icon on one of the toolbars.



You can place several instructions on a rung in sequence by clicking the icons one after another. RSLogix 500 places instructions from left to right.

RSLogix 500 supports a file-based editor. This means that you can:

- create and/or edit multiple rungs at a time
- enter addresses before you actually create data table files for your I/O

- enter symbols before you have assigned addresses for them in the database
- enter instructions without having to provide addresses until just before file validation occurs

To add addresses just click an instruction and then type the address in the empty field that appears above the instruction. With **RSLogix 500** you can also drag and drop addresses from a data table file onto instructions in your ladder logic.

Remember to use the right mouse button to access functionality whenever possible. The right mouse button provides you with context menus that list editing options. And always remember that you can click F1 (or the Help button when available) on any instruction, or within any window to access help.

Keyboard users can press the [Shift + F10] key combination to access a right mouse menu.

**TIP**

You can select multiple rungs by holding down the [Ctrl] key and clicking the left mouse button on every rung you want to select. You can also select a range of rungs by holding down the [Shift] key and clicking the beginning rung and ending rung.

When you select rungs in this manner, **RSLogix 500** remembers the order in which you made your selections, and pastes the rungs to the clipboard in that order. When you paste the rungs, the order in which you copied them is retained. For example if you click rung 11 and then [Shift] click rung 8 to copy a range of rungs, the rungs are copied to the clipboard from rung 11 to rung 8. Pasting these rungs will place them in the new location in this same order.

## Step 6 ■ Add documentation to your logic instructions

You can use several methods to add symbols and descriptions to addresses in the database.

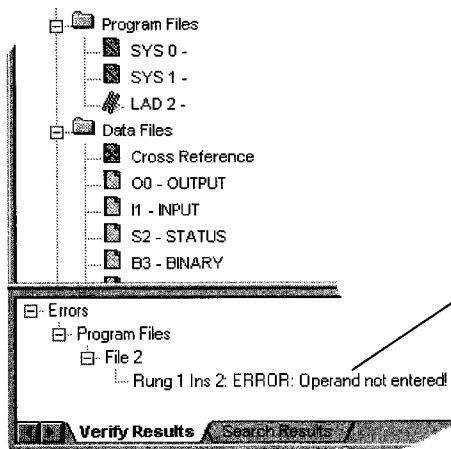
- Open the program file and add the documentation directly to the addressed instruction. Use the right mouse menu for this.
- Modify an address's assigned documentation in the data file. Double-click the data file in the project tree, and then click on an address within the grid that appears on the data file dialog. At the bottom of the dialog there are fields where you can enter the documentation for the address.

- Modify the database using the database editor. Double-click an icon in the database folder located in the project tree.
- Enter a symbol directly and later assign an address to the symbol using the database symbol/description editor.

## Step 7 ■ Verify the program logic

When you are ready to build your project, you can validate a single program file or you can validate your entire project. Use the menu bar or the right mouse button menu to initiate this process.

- After you initiate a verification, the Verify Results output window displays and gives you information about mistakes or omissions that may have occurred as you wrote your program logic.



The results of any verification are displayed at the bottom of the window under the project tree. To hide this results window after viewing it, click the View menu and then click Results.

## Step 8 ■ Configure communication channel, download, and go online

### TIP



If you are developing the program offline, for example on a laptop remote from the site, and later plan to download and run the program on a specific processor (node) via a determined protocol, you may want to override system communication settings made in step 1. Do this from the Controller Properties window, Controller Communications tab. Settings made via this method will override any driver and node settings established in step 1, and should be completed before performing this step.

Before going online you have to define processor communication settings such as baud rate, and also decide certain system and protocol controls. Depending on the type of processor that you are using and the method of communication (direct vs. networked or modem), the complexity of this procedure varies.

Double-click the channel configuration icon in the project tree to make these settings. If you need information about any parameter, click Help on the channel configuration window.

Finally, from the Comms menu, click Download to download the current offline program into the controller. **RSLogix500** will ask if you want to go online. Click yes to go online; then select the operating mode.

## Step 9 ■ Monitor data files

You can use **RSLogix 500** to watch what is happening in your data table files. This procedure is called monitoring data table files.

When you are monitoring data table files, you can:

- define how your data file selection grid will display
- change values in the data table
- change the display radix
- show which addresses are used in your ladder logic
- switch between files
- quickly jump to another address in another data table file

Double-click the data file icon in the project tree that contains the data you want to monitor. You can have multiple data table files opened for monitoring at the same time. Just drag each data table window into viewing position by clicking on the title bar and moving the mouse. Release the mouse button to place the data table window.

You can also choose to cascade or tile all the windows opened in your **RSLogix** project by selecting a viewing option from the Window menu.

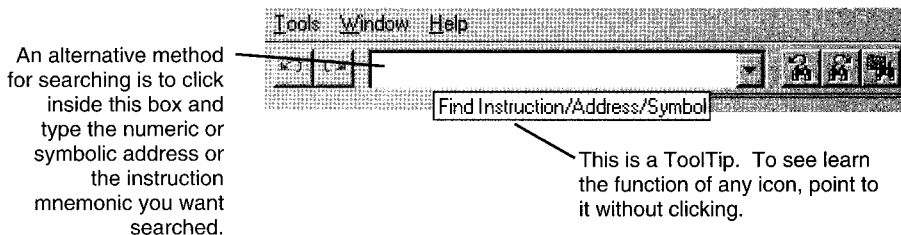
Data changes made offline only affect the disk file unless the program is restored to the processor.

Data changes made online only affect the processor file unless the program is saved or uploaded while online to update the disk file.

## Step 10 ■ Search and replace instructions

The Find option allows you to quickly locate instructions, addresses, and symbols (if they have been defined) in ladder program files. You can even search for edit zones within your logic program. If you want to automatically replace instructions and addresses with different ones, you can use the Replace option. You can use wildcards in your find and replace operations.

Begin any find or replace operation by selecting that functionality from the Search menu. Then type the mnemonic (XIC, TON, etc.), the address (B3/4, etc.) or a combination of both mnemonic and address (XIC B3/4) or mnemonic and symbol (XIC REPEAT) for the instruction you want to locate in the Find What text box.



## Step 11 ■ Print a report

When you need a printed copy of what you've done in RSLogix 500, you need to print reports. There are many reports available and you can select which reports you want to print from the Report Options dialog. Select this dialog from the File menu.

### TIP



To preview the way a ladder file will print, click Preview. You can scale up the image to make the instructions appear larger on the printed page or scale down the image so that, in cases where many instructions are on a rung of logic, all the instructions can fit on the printed page.

This chapter provides information that you can use to make editing your ladder logic easier.

### TIP



Shortcut methods exist for most editing functions within RSLogix 500. You can access this list of shortcuts in the online help by searching the word “shortcuts” in the online help.



## Backing up your work

Remember to back up your work as you develop your ladder logic programs.

RSLogix 500 uses two types of backup files that you can access at any time, and provides you with an auto-recovery file in the case of a power failure. All of these files contain the entire description database associated with the project..

- Auto-Backup files are created automatically each time you save a project. You can preset how many backups should be retained for any project by entering a Number of Backups in the Preferences tab of the System Options dialog. Reach this tab from the Tools menu. Then click Options and select the Preference tab. Auto-backup files (saved as .RSS files) have the letters BAK and a series of numbers (000 to 999) appended to the filename. For example, an auto-backup created for project TEST.RSS might be identified as TEST\_BAK000.RSS, and a more recent backup might be identified as TEST\_BAK001.RSS.
- Compressed Format Backup files are typically generated for archiving or giving to another user. Compressed format backup files include the .RSS and all database files for the project compressed into a single .RS1 file. From the File menu click Backup Project to generate a compressed-format backup file.

## Crash recovery

If you experience a power interruption, RSLogix 500 provides you with a recent backup file containing current edits.

RSLogix 500 automatically creates file backups while you are working with a project and when you save the project. This auto-generated recovery file (internal RSS file) is only available to you the next time you open a project if you have a system crash or your power is interrupted. After attempting to open a project after a power failure RSLogix 500 prompts you with choices.

You can open:

- the auto-saved file, ensuring retention of any edits made before the power interruption.
- the last backup that you made, when you selected Save before the power interruption.

---

**IMPORTANT**

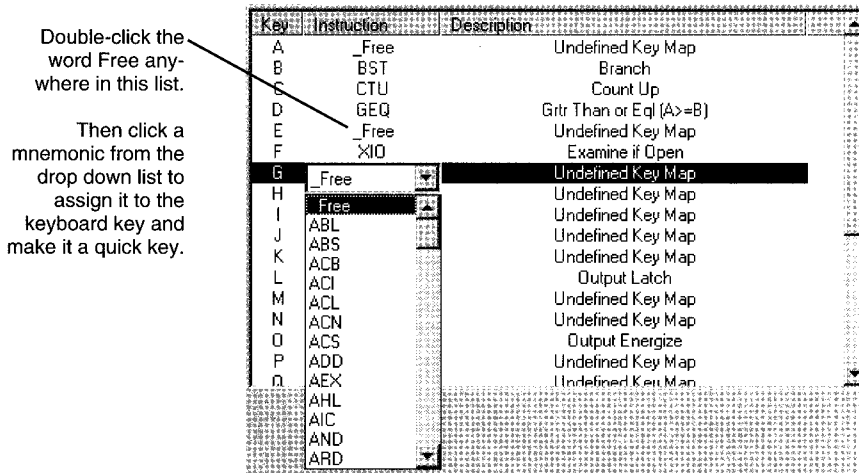
You must have saved or closed the file you are working on at least one time for the auto-recovery process to work. Therefore, it is good practice to save the file immediately after beginning a new project. This ensures that your auto-recovery process can begin properly.

---

You can set the interval time at which auto-recovery saves of your project will occur. Do this by making a setting in the Preferences dialog. The auto-recovery process ensures that you will be able to retain any work that had been done on the file between the time of the power interruption and the last manual save.

## Quick entry of instructions

To make your programming tasks faster, RSLogix 500 lets you map any available alphabetic key (A-Z) on your computer keyboard to a ladder logic programming instruction.



From the View menu, click Properties. Then click Quick Key Mapping to access the mapping list. Make sure you have a program file window opened and active or you will not be able to select Properties from the View menu.

### TIP



You can jump to any rung in your project by clicking the Search menu, and then clicking Goto. You can go to a rung in the current program file or you can go to a rung in another program file within the same project.

Keyboard users can press the [Ctrl + G] key combination to access the Goto Rung dialog.

## Addressing

There are several different methods that you can use to address instructions. You can enter an address by:

- manually typing it in
- dragging addresses from data files
- using copy and paste from program to program

**TIP**

You can drag-and-drop rungs, branches, instructions, and addresses from file to file or from the database to a file. To drag-and-drop, position the mouse pointer over a file element, click and hold down the left mouse button and drag the element to another location, and then release the mouse button. Red boxes indicate valid locations; these turn green when properly selected.

## Branching

### Add a branch



Click this icon on the instruction toolbar to place a branch in your ladder logic. If your cursor is on an instruction, the branch is placed immediately to the right of the instruction. If your cursor is on the rung number, the branch is placed first on the rung.

### Move a branch



Click on the upper left corner of a branch to move the entire branch structure to another location in your ladder logic program.

### Expand a branch



Click the right leg of the branch, then drag the leg to the right or left. Valid release points will be visible on the ladder display.

### Nested branches



Place the cursor at the upper left corner of a branch leg, click the right mouse button, and select Append New Branch to place another branch structure within the original branch structure.

### Parallel branches



Place the cursor at the bottom left corner of a branch leg and click the right mouse button to Extend Branch Leg Up or Extend Branch Leg Down.

## Copy branch leg



Click on the left edge of the branch leg you want to copy. In the picture at the left this is the center leg. Then click copy in the right mouse menu. Finally click on a rung or instruction in your logic and click paste from the right mouse menu to insert the rung leg

## Copy entire branch structure



Select the right leg of the branch structure, then click copy in the right mouse menu. Finally click on a rung or instruction in your logic and click paste from the right mouse menu to insert the rung structure.

## Delete a branch

Place the cursor at any location on the branch and click the right mouse button. Then click Delete. If you cut or delete a branch, all instructions on the branch are also deleted.

## Branching restrictions

You are limited to a maximum of 75 parallel branches.

You are limited to a maximum of 4 nested branches. (SLC 5/02 and higher).

## Undo operation



The undo icon reverses your last action. You can use this icon to walk through (and undo) your previous actions one at a time. RSLogix 500 remembers up to 200 previous actions.

If you want to undo a move operation, you must click undo two times. This is because RSLogix 500 considers a move a series of two actions (copy and cut). You have to let RSLogix know that you want both the copy and the cut undone. If you click undo only one time when trying to undo a move, the move appears to be a copy, and you will see the moved element appear at both locations.

## Online editing

The online editing function lets you monitor and correct your ladder program when your programming terminal is connected to a SLC 5/03, SLC 5/04 or SLC 5/05 processor. Only one programming device at a time can perform online edits of the program.

Online editing functions consist of inserting, replacing, and deleting rungs in an existing ladder program while online with the processor.

Within your logic program **RSLogix 500** places zone markers in the margin to the left of the left rail. These letters signify edit zones and they indicate the type of ladder program edit that exists in the program.

Lower case zone markers indicate edits that exist only in the computer memory. Upper case zone markers indicate edits that exist in the processor memory. After successfully assembling the edited rungs, the zone markers disappear.

---

**TIP**

You can search for zone markers in your project the same way you might search for an instruction or an address. Do this via the Find dialog using the Special button.

---

## Lower-case zone markers

- e** (Offline and online, all controllers) These rungs are currently under edit within the computer RAM. If you are working offline, after a successful program verification the lower-case e will disappear and the edits will be incorporated into the program. If you are working online, after accepting the rung, the lower-case e will be replaced by an upper-case I indicating that the rung is now in the controller's memory and will be inserted into the program file.
- i** (Online Editing, SLC 5/03, 5/04 and 5/05 controllers only) These rungs are to be inserted into the program. Rungs marked with a lower-case i currently exist in the computer memory and will not be entered into the controller until the rung is accepted (right mouse button selection). After the rung is accepted, the lower-case i is replaced by an upper-case I.
- r** (Online Editing, SLC 5/03, 5/04 and 5/05 controllers only) These rungs are to be replaced in the ladder program. Rungs marked with a lower-case r currently exist in the computer memory and will not be entered into the controller until the rung is accepted (right mouse button selection). An r marked rung is always preceded by an e marked rung. After the rung is accepted, the lower-case r will be replaced by an upper-case R.
- d** (Online Editing, SLC 5/03, 5/04 and 5/05 controllers only) These rungs are to be deleted from the ladder program. Rungs marked with a lower-case d indicate a deletion reflected in the computer memory. This deletion will not be reflected in the controller until the rung is accepted (right mouse button selection), at which time it will be replaced by an upper-case D.

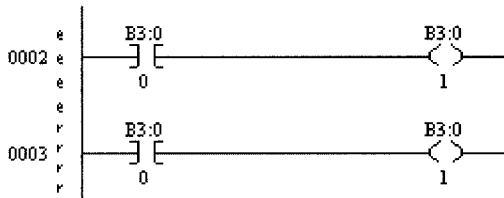
## Upper-case zone markers

- I** (Online Editing, SLC 5/03, 5/04 and 5/05 controllers only) These rungs have been inserted in the controller's logic program. You can test the edits by selecting the Edit menu and clicking Test Edits to see how the rung works in the online ladder program. Click Assemble Edits to finalize the rung insertion and complete the editing process.
- R** (Online Editing, SLC 5/03, 5/04 and 5/05 controllers only) These rungs have been replaced in the controller's logic program. Rungs marked with an upper-case R continue to function in the program until you select Test Edits to see how the new rung works in the online program.. Select Assemble Edits to finalize the replacement and complete the editing process.
- D** (Online Editing, SLC 5/03, 5/04 and 5/05 controllers only) These rungs have been deleted in the controller's logic program. Rungs marked with an upper-case D continue to function in the program until you select Test Edits to see how the program functions without the rungs in the online program. Select Assemble Edits to finalize the deletion and complete the editing process.

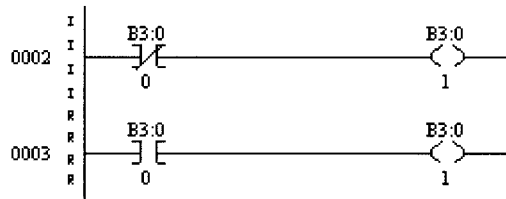
## Online editing example

This example replaces an XIC instruction with an XIO instruction with the same address while online.

1. Select the rung in the program that requires editing and then from the Edit menu, select Start Rung Edits from the main menu or choose Start Rung Edits from the right mouse menu. A duplicate of the selected rung (preceded by the **e** edit zone marker) is shown in your program. This is the rung that all edits will be performed on. The **r** edit zone marker precedes the original rung (rung to be replaced). See the example below.



2. Make the edits to the rung. The lower-case edit markers do not change since they represent changes that only exist in the computer memory; these changes are not yet a part of the online program in the controller. (At this step you can click Cancel Rung Edits to cancel the edits you have made to the rung.)
4. Select Accept Rung. This changes the edit zone markers and places both rungs in the controller memory. The upper-case **I** represents the rung that has been inserted into the online program. The upper-case **R** represents the online rung that is to be replaced. At this time the **R** rung is still operating in the program.



5. Select Test Edits. The **I**-marked rung takes precedence. The program in the controller will operate with the inserted rung, and the **R**-marked rung will be ignored. (Alternately you can click Cancel Edits to cancel the accepted **I**-marked rung and retain the originally programmed **R**-marked rung instead.)
6. Select Assemble Edits. All edit zone markers disappear and the edits are incorporated into the online program. There is no Undo option after online edits have been assembled.

Going from online to offline with rungs under edit removes the RAM online edits. Make sure you have accepted edits before going offline if you want any changes retained in the processor.

## Online editing restrictions

Your programming terminal must be connected to a SLC 5/03, SLC 5/04 or SLC 5/05 processor. During an online editing session you cannot:

- resize data table files
- create or delete program files
- change program file protection
- change index across file boundaries flag
- reconfigure the I/O
- select force protection

## ASCII editing

ASCII Editing is a function of **RSLogix 500** that lets you modify instructions using ASCII instruction mnemonics instead of having to modify instructions using the ladder editor.

A quick way to call the ASCII Editor is to double-click a rung number in the left margin. If you double-click a rung with logic already on it, you will see the mnemonics for the existing instructions and can modify or add to them. If you double-click an empty rung, you get an empty editing box into which you can type the mnemonics that represent the logic you want placed on the rung.

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**TIP**

Another quick method to call the ASCII text editor is to click the rung number and then press the forward slash key (/) on your keyboard.

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## Configuring Interrupts

### STI (Selectable Timed Interrupt)

You configure STIs on the status dialog. STIs are programmed to execute at predetermined intervals.

Select a program file for the STI by double-clicking the S2 data file icon in the project tree. Then click the STI tab and enter the information needed to define the STI. Press the Help button if you need more information.

### DII (Discrete Input Interrupt)

You configure DIIs on the status dialog. Use the DII for high-speed processing applications or any application that needs to respond to an event quickly. This function allows the processor to execute a ladder subroutine when the input bit pattern of a discrete I/O card matches a compare value that you programmed.

Double-click the S2 data file icon in the project tree. Then click the DII tab and enter the information needed to define the DII. Press the Help button if you need more information.



# APPENDIX C

## Glossary

**absolute optical encoder** An optical rotary encoder that outputs a binary word representing the angular position.

**absolute pressure** The pressure difference between the measured value and an absolute vacuum.

**AC servomotor** A two-phase motor used in control applications; designed to have a near-linear torque-speed curve.

**acceleration** The process of increasing velocity; uniform acceleration is when the velocity is increasing at a constant rate.

**accumulator** A temporary, digital data-storage register in the microprocessor used in many math, logic, and data-moving operations. Used in a hydraulic system, a spring-loaded tank connected into the high-pressure side of the system; the accumulator can store some fluid to be used at peak-demand periods.

**active filter** A circuit that incorporates an op-amp.

**actuator** The source of energy or motion in a system; typical actuators would be electric motors or hydraulic cylinders.

**adaptive controller** A controller, such as a PID, with the ability to monitor and improve its own performance.

**ADC** *See* **analog-to-digital converter**.

**address** A number that represents the location of 1 byte of data in memory or a specific input/output port.

**address bus** A group of signals coming from the microprocessor to memory and I/O ports, specifying the address.

**aftercooler** Used in pneumatic systems, a device that removes moisture from the air supply by chilling and thereby condensing any water vapor.

**air compressor** Used in pneumatic systems, a device that creates the supply of high-pressure air.

**aliasing** The event that occurs when the wrong data are reconstructed from the sampled sensor data because the sample rate is too slow.

**ALU** *See* **arithmetic logic unit**.

**analog drive** A method of controlling an electric motor's speed by varying the DC supply voltage.

**analog I/O module** A module that plugs into the PLC and converts real-world analog signals to digital signals for the PLC and vice versa.

**analog switch** A solid-state device that performs the same function as a low-power mechanical switch.

**analog-to-digital converter (ADC)** A device (usually an IC) that can convert an analog voltage into its digital binary equivalent.

**anode** One of three terminals of an SCR and PUT.

**arithmetic logic unit (ALU)** The part of the CPU that performs arithmetic and logical operations.

**armature** The part of a motor that responds to the magnetic field; typically, the armature is the rotating assembly of an electric motor.

**assembly language** A computer program written in mnemonics, which are English-like abbreviations for machine-code instructions.

**autotuning** The ability of some digital controllers to continuously fine-tune themselves by making small changes in their gain constants, based on past performance.

**axial piston pump** A type of hydraulic pump that uses a number of small pistons to pump the fluid, this is a variable-displacement pump because the stroke of the pistons is determined by the adjustable angle of the swash plate.

**backlash** In a gear pass, the small amount of free play between gears so that the teeth don't bind on each other.

**ballast resistor** A resistor placed in series with the stepper motor coils to improve the torque at higher stepping rates by reducing the motor-time constant.

**ball bearing screw** A ball bearing type of a lead-screw, recirculating balls roll between the threaded shaft and the nut to remove almost all friction.

**band-pass filter** A circuit that allows only a specified range of frequencies to pass and attenuates all others above and below the pass band.

**base** One of three terminals of a transistor.

**batch process** A process that has a beginning and an end and is usually preformed over and over.

**baud** The rate at which the signal states are changing; frequently used to mean "bits per second."

**bias** A value that is added to the output of the controller to compensate for a constant disturbing force, such as gravity.

**biasing circuit** The part of a transistor circuit that generates the forward bias voltage.

**bidirectional control valve** Used in hydraulic and pneumatic systems, it can direct the fluid into either end

of the actuator cylinder, allowing the piston to be moved in either direction.

**bilevel drive** A technique that uses two voltages to improve torque at higher stepping rates. A higher voltage is applied to the motor at the beginning of the step, and then a lower voltage is switched in.

**bimetallic temperature sensor** A strip made from two metals with different coefficients of thermal expansion; as the strip heats up, it bends.

**bipolar junction transistor (BJT)** Known as a *transistor*, a solid-state device that can be used as a switch or an amplifier.

**bipolar motor** A stepper motor that requires polarity reversals for some of the steps.

**bit** The smallest unit of digital data, which has a value of 1 or 0.

**bit instruction** The basic PLC instruction that logically represents a simple switch or relay coil; the state of each switch or coil occupies 1 bit in memory.

**BJT** See **bipolar junction transistor**.

**BLDC** See **brushless DC motor**.

**Bode plot** A graph that plots system gain and phase shift versus frequency; it can be used to determine system stability.

**bonded-wire strain gauge** A force sensor consisting of a small pattern of thin wires attached to some structural member; when the member is stressed, the stretched wires slightly increase the resistance.

**Bourdon tube** A type of pressure sensor consisting of a short bent tube closed at one end; pressure inside the tube tends to straighten it.

**brake** A device used to slow down or stop a shaft, or to prevent a stopped shaft from moving.

**brush** A stationary conductive rod that rubs on the rotating commutator and conducts current into the armature windings.

**brushless DC motor (BLDC)** The newest type of DC motor; it does not use a commutator or brushes; instead, the field windings are turned on and off in sequence, and the permanent magnet rotor (armature) is pulled around by the apparently rotating magnetic field.

**bus topology** A type of network wherein all the nodes tap into a single cable at different places, and the cable has a beginning and an end (that is, the cable is not in a ring).

**byte** An 8-bit digital word.

**CAD** *See* **computer-aided design**.

**CAM** *See* **computer-aided manufacturing**.

**capacitor start motor** A single-phase AC induction motor that uses a capacitor to phase-shift the AC for the start windings.

**cathode** One of three terminals of an SCR and PUT.

**CEMF** *See* **counter-EMF**.

**central processing unit (CPU)** The central part of a computer, the CPU performs all calculations and handles the control functions of the computer.

**change-of-state system** A method of communicating on a network wherein a device sends a message only when something changes.

**chatter** the condition that occurs when the output of a comparator oscillates when the input voltage is near the threshold voltage.

**CIM** *See* **computer-integrated manufacturing**.

**circular pitch** The distance along the pitch circle of a gear of one tooth and valley.

**class A operation** The event when a transistor is biased so the collector current is approximately half of its maximum value.

**class B operation** The event when a transistor is biased so that the collector current is just turning on.

**class C operation** The event when a transistor is biased below cutoff and used as an on-off switch.

**closed-loop control system** A control system that uses feedback. A sensor continually monitors the output of the system and sends a signal to the controller, which makes adjustments to keep the output within specification.

**closed-loop gain** The gain of an amplifier when feedback is being used. The value of closed-loop gain is less than open-loop gain (for negative feedback).

**closed-loop control system** A control system that uses feedback. A sensor continually monitors the output of the system and sends a signal to the controller, which makes adjustments to keep the output within specification.

**clutch** A device used to connect or disconnect rotating shafts “on the fly”; usually a clutch acts by pressing two friction disks together.

**coefficient of friction** A constant that is arrived at experimentally for different materials and can be used to calculate the force needed to overcome friction.

**cogging** The jerky motion of electric linear motors, especially at low speeds.

**cold junction** One of two temperature-sensitive junctions of the thermocouple temperature sensor. The cold junction is usually kept at a reference temperature.

**collector** One of the three terminals of a transistor.

**common mode rejection** System whereby a common voltage in both wires of a differential signal is canceled out. This system is used to advantage to reduce noise that is usually common to both wires.

**commutator** The part of an armature that makes contact with the brushes.

**comparator** (1) A type of op-amp that is used open-loop to determine if one voltage is higher or lower than another voltage. (2) Part of the control system that subtracts the feedback signal (as reported by the sensor) from the set point, to determine the error.

**comparison instruction** Compares two words and yields a single bit as a result.

**compensating gauge** A nonactive strain gauge that is used solely for the purpose of canceling out temperature effects.

**compound motor** A motor that has both shunt and series field windings.

**computer-aided design** A computer system that makes engineering drawings.

**computer-aided manufacturing** A computer system that allows CAD drawings to be converted for use by a numerical control (NC) machine tool.

**computer-integrated manufacturing** A computer system that oversees every step in the manufacturing process, from customer order to delivery of finished parts.

**constant current chopper drive** A drive circuit for stepper motors that uses PWM techniques to maintain a constant average current at all speeds.

**contactor** A heavy-duty relay that switches power directly to motors and machinery.

**continuous-cycle method** A method of tuning a PID controller that involves driving the controller into oscillations and measuring the response; the PID constants are computed based on these data.

**continuous-level detector** A sensor that can determine the fluid level in a container.

**continuous process** A process wherein there is a continuous flow of product—for example, a steam boiler where water is continuously pumped in and steam continuously comes out.

**control bus** A group of timing and control signals coming from the microprocessor to memory and I/O ports.

**control instruction** Allows for such things as jumps and subroutines (in a computer program).

**control strategy** The set of rules that the controller follows to determine its output to the actuator.

**control system** A system that may include electronic and mechanical components, where some type of machine intelligence controls a physical process.

**control valve** Used to control the flow of fluids. A valve that has a built-in valve-operating mechanism, allowing it to be controlled remotely by a signal from the controller.

**control winding** One of two windings in an AC servomotor.

**controlled variable** The ultimate output of the process; the actual parameter of the process that is being controlled, such as temperature in a heating system.

**controller** The machine intelligence of the control system.

**counter-EMF (CEMF)** A voltage that is generated inside an electric motor when running under its own power; the polarity of the CEMF is always opposite to the applied voltage.

**CPU** *See* **central processing unit**.

**creep** The property that explains why two pulleys connected with a belt will not stay exactly synchronized.

**critically damped** A system that is damped just enough to prevent overshoot.

**cumulative compound motor** A compound motor where the magnetic fields of the shunt and series fields aid each other.

**cumulative error** Error that accumulates; for example, a cumulative error of  $1^\circ$  per revolution means that the measurement error would be  $5^\circ$  after  $5^\circ$  revolutions.

**current loop** In a signal transmission system, a single loop of wire that goes from the transmitter to the receiver and back to the transmitter. The signal intelligence is conveyed by the current level instead of the

voltage. This system is immune to voltage drops caused by wire resistance.

**current-to-voltage converter** An op-amp based circuit used as a receiver for a current-loop system.

**cutoff frequency** The frequency at which a filter circuit starts to attenuate.

**DAC** *See* **digital-to-analog converter**.

**damping** A drag force on a system from sliding or viscous friction, which acts to slow movement (make it sluggish).

**data bus** A group of signals going to and from the microprocessor, memory, and I/O ports. The data bus carries the actual data that are being processed.

**data communication equipment (DCE)** One of two units specified by the RS-232 standard (for serial data transfer); the DCE is usually a modem.

**data file** The collection of data in a PLC that represents the present condition of switches, counters, relays, and the like.

**data memory** The section of RAM memory in a PLC where the data file is stored.

**data terminal equipment (DTE)** One of two units specified by the RS-232 standard (for serial data transfer). The DTE is usually the computer.

**data packet** A message transmitted on a network; usually includes the address of where it's going, the address of where it came from, and the message itself.

**DCC** *See* **distributed computer control**.

**DCE** *See* **data communication equipment**.

**DC link converter** A circuit that converts line voltage at 60 Hz to AC power at different voltage and frequencies, for the purpose of controlling the speed of AC motors.

**DC offset voltage** The small voltage that may occur on the output of an op-amp, even when the inputs are

equal; DC offset can be eliminated with a resistor adjustment.

**DDC** *See* **direct digital control**.

**dead band** The small range, on either side of the set point (of the controlled variable), where the error cannot be corrected by the controller because of friction, mechanical backlash, and the like.

**dead time** The interval of time between the correcting signal sent by the controller and the response of the system.

**dead zone** *See* **dead band**.

**decibel** Gain expressed in a logarithmic scale.

**delta-connection** One of two ways in which to connect three-phase motor or generator coils. With the delta-connection, each of the three coils is connected between two phase wires; there is no neutral.

**derivative control** A feedback control strategy where the restoring force is proportional to the rate that the error is changing (increasing or decreasing); derivative control tends to quicken the response and reduce overshoot.

**desiccant dryer** Used in pneumatic systems, a device that dries the air supply by passing the air through a water-absorbing chemical.

**detent torque** A magnetic tug that keeps a stepper motor rotor from turning even when the power is off; also called *residual torque*.

**deterministic network** A network able to assign priority to a message and to guarantee that it will arrive in a given time.

**device-level network** A network used to connect low-level devices such as sensors and actuators to a PLC.

**DeviceNet** A type of device-level network developed by Allen-Bradley. *See* **device-level network**.

**diac** A bistable, two-terminal, solid-state device that is used to trigger the triac.

**diametral pitch** The ratio of the number of gear teeth per inch of pitch diameter; in practice, the number that describes the tooth size when specifying gears.

**differential amplifier** A circuit that produces an output voltage that is proportional to the instantaneous voltage difference between two input signals. Op-amps by themselves are difference amplifiers; however, a practical differential amplifier incorporates additional components.

**differential compound motor** A compound motor where the magnetic fields of the shunt and series windings oppose each other.

**differential pressure** The difference between two pressures where neither may be ambient.

**differential voltage** A signal voltage carried on two wires, where neither wire is at ground potential.

**differentiator** An op-amp circuit that has an output voltage proportional to the instantaneous rate of change of the input voltage.

**digital control system** A control system where the controller is a digital circuit, typically a computer.

**Digital-signal processor (DSP)** A type of digital processor that is designed to support high-performance, repetitive, numerically intensive tasks.

**digital-to-analog converter (DAC)** A circuit that translates digital data into an analog voltage.

**DIP switch** A set of SPST switches built into the shape of an IC (DIP stands for dual in-line package).

**direct current tachometer** Essentially, a DC generator that gives an output voltage proportional to angular velocity.

**direct digital control** An approach to process control where all controllers in a large process are simulated by a single computer.

**discrete I/O** I/O that deals with on-off devices. Discrete input modules are used to connect real-world switches to the PLC, and discrete output modules are used to turn on lamps, relays, motors, and so on.

**discrete-level detector** A sensor that can determine if the fluid in a container has reached a certain level.

**distributed computer control** An approach to process control where each process has its own local controller, but all individual controllers are connected to a single computer for programming and monitoring.

**double-acting cylinder** A hydraulic or pneumatic cylinder and piston assembly in which the piston can be driven in either direction.

**double-pole/double-throw switch (DPDT)** A switch contact configuration.

**download** To transfer a computer program or data into a computer (from another computer).

**DPDT** See **double-pole/double-throw switch**.

**drain** One of three terminals in an FET.

**dryer** Used in pneumatic systems, a dryer removes the moisture from the air supply.

**DTE** See **data terminal equipment**.

**duty cycle** The percentage of cycle time that the PWM pulse is high—that is, a true square wave has a duty cycle of 50%.

**dv/dt effect** The event when the SCR turns on if the anode-cathode voltage rises too quickly.

**dynamic braking** A method of braking the motor by switching the armature windings to a resistor. The coasting motor generates current that is dissipated in the resistor, thus removing energy from the system.

**dynamic torque** The motor torque available to rotate the load under normal conditions.

**earth ground** The voltage at (or connection to) the surface of the earth at some particular place.

**efficiency** When describing energy conversions, the percentage of input energy that is converted to useful output energy.

**eight-step drive** A two- or four-phase stepper motor being driven in half-steps; the sequencing pattern has eight steps.

**electric field** A condition that exists in the space between two objects that are at a different voltage potential. A wire in this space will assume a “noise” voltage proportional to the field strength.

**electromechanical counter** A device that counts events. Each time it receives a pulse, it increments a mechanical counter, which can then be read by an operator or used to activate some other process.

**electromechanical relay (EMR)** A device that uses an electromagnet to close (or open) switch contacts—in other words, an electrically powered switch.

**electromechanical sequencer** A device that provides sequential activation signals; it works by rotating a drum with cams, which activates switches.

**embedded controller** A small microprocessor-based controller that is permanently installed within the machine it is controlling.

**emitter** One of three terminals of a transistor.

**EMR** *See* **electromechanical relay**.

**energy** The amount of work it takes to do a job; energy has different units for chemical, thermal, mechanical, and electrical systems.

**enhancement mode** Property of some MOSFETs where the gate voltage can always be positive (for N-channel).

**error (E)** In a control system, the difference between where the controlled variable should be (set point) and where it actually is ( $E = SP - PV$ ).

**Ethernet** The type of network used to interconnect PCs and mainframe computers for the purpose of exchanging data and data files.

**event control system** A control system that cycles through a predetermined series of steps.

**event-driven operation** In a sequentially controlled system, an action that is allowed to start or continue based on some parameter changes. This is an example of closed-loop control.

**expansion card/slot** An expansion card is a printed circuit card that plugs into an expansion slot on the motherboard of a personal computer (PC). The expansion card usually interfaces the PC to the outside world.

**extension rate** The rate at which a linear actuator can extend or retract; the term is usually applied to electrical linear actuators.

**feedback** The signal from the sensor, which is fed back to the controller.

**feedforward** The concept of letting the controller know in advance that a change is coming.

**fetch-execute cycle** A computer cycle where the CPU fetches an instruction and then executes it.

**FET** *See* **field effect transistor**.

**field** The part of an electric motor that provides a magnetic field; typically, it is the stationary part.

**Fieldbus** A type of device-level network. *See* **device-level network**

**field effect transistor (FET)** A three-terminal solid-state amplifying device that uses voltage as its input signal.

**field-oriented control drive** A class of AC-motor-control units that provides very precise control of speed. A type of vector drive. *See* **vector drives**.

**field winding** A stationary electromagnet used to provide the magnetic field needed by the armature.

**flow sensor** A sensor that measures the quantity of fluid flowing in a pipe or channel.

**flow-control valve** Used to control the flow of fluids. A valve that has a built-in valve-operating mechanism,

allowing it to be controlled remotely by a signal from the controller.

**flux vector drive** A class of AC-motor-control units that requires a position sensor and maintains better control than the sensorless drive can (but not as good control as the field-oriented drive). A type of vector drive. See **vector drives**.

**follow-up system** A control system where the output follows a specified path.

**forced commutation** The process of turning off an SCR by momentarily forcing the anode-cathode voltage to 0 V (or below).

**forward bias voltage** The DC offset base voltage required to start the transistor conducting.

**forward breakover voltage** The voltage across a thyristor that causes it to switch into its conduction state.

**forward conduction region** The operating range of a thyristor when it is conducting.

**forward current gain ( $h_{FE}$ )** The gain of a transistor; the collector current divided by the base current.

**forward path** The signal-flow direction of the controller to the actuator.

**four-phase stepper motor** A motor with four separate field circuits; this motor does not require polarity reversals to operate and hence is unipolar.

**four-step drive** The standard operating mode for two- or four-phase stepper motors taking full steps; the sequencing pattern has four states.

**fractional horsepower (hp) motor** A motor with less than 1 hp.

**frame** A basic part of an AC motor, an overall case that supports the stator and the rotor bearings.

**fuzzy logic** A new control strategy, modeled originally on human thought processes, where decisions are made based on a number of factors and each fac-

tor can have such qualities as *maybe* and *almost*, to name only two.

**fuzzy predicates** Linguistic terms of degree, used in describing actual variables—for example, *warm*, *very warm*, *not so warm*, and so on.

**fuzzy set** The range of one fuzzy predicate in a system—that is, all values of temperature that correspond to *very warm*.

**gain** A multiplication factor; the steady-state relationship between input and output of a component. (In this text, *gain* and *transfer function* are used interchangeably, although this is a simplification.) The term *system gain* refers to the proportional gain factor ( $K_p$ ).

**gain-bandwidth product ( $f_T$ )** A constant for an amplifier that is the product of the open-loop gain and the frequency at that gain; can be read as the frequency when a transistor's gain has been reduced to 1.

**gain margin** A form of a safety margin. When the phase lag is  $180^\circ$ , the gain margin is the amount the gain could be increased before the system becomes unstable.

**gate** One of three terminals in an SCR, PUT, FET, or triac.

**gauge pressure** The difference between measured and ambient pressure (ambient is 14.7 psi or 101.3 kPa).

**gearhead** In smaller motors, a gear train attached or built-in to the motor assembly, which effectively gives the motor more torque at less rpm.

**gear motor** A simple hydraulic motor in which fluid pushes on the teeth of a set of constantly meshed gears (in a housing), causing them to rotate.

**gear pass** Two gears in mesh.

**gear pump** A simple positive-displacement hydraulic fluid pump that uses two constantly meshed gears (in a housing).

**gear ratio** The ratio of the number of teeth of two gears in mesh. (Also the ratio of pitch diameters.)



**gear train** A gear system consisting of more than one gear pass.

**GFI** *See* **ground-fault interrupter**.

**Grey code** A sequence of digital states that has been designed so that only 1 bit changes between any two adjacent states.

**ground-fault interrupter (GFI)** A safety device that will shut off the power if it senses a current leakage to the safety ground.

**half-steps** For stepper motors, by alternating the standard mode with the dual excitation mode, the angle of step will be half of what it normally is.

**Hall effect** The phenomena of a semiconductor material generating a voltage when in the presence of a magnetic field; used primarily as a proximity sensor.

**harmonic drive** A unique device using a flexible gear that can provide a large gear ratio with virtually no backlash.

**head** The fluid pressure in a tank, which is caused by the weight of the fluid above and is therefore proportional to the level.

**heat sink** A piece of metal, possibly with cooling fins, used to dissipate heat from a power device to the air.

**high-pass filter** A circuit that allows higher frequencies to pass but attenuates lower-frequency signals.

**holding current ( $I_H$ )** Once the SCR has been turned on, the small current (to the SCR) necessary to keep the SCR in the conduction state.

**holding torque** The stepper motor torque available to keep the shaft from rotating when the motor is stopped but with the last field coil still energized.

**Hooke's law** The “spring law” that states the deformation in a spring is directly proportional to the force on the spring.

**hot junction** One of two temperature-sensitive junctions of the thermocouple temperature sensor. The hot junction is used on the probe.

**hybrid solid-state relay** A device that uses a reed relay to activate a triac.

**hybrid stepper motor** A motor that combines the features of the PM and VR stepper motors—that is, it can take small steps and has a detent torque.

**hydraulic cylinder** A linear actuator powered by fluid under pressure, it consists of a cylinder, enclosed at both ends with a piston inside. The piston is moved by admitting the pressurized fluid into either end of the cylinder.

**hydraulic system** A system that uses pressurized fluid (oil) to power actuators such as hydraulic cylinders.

**hydrostatic pressure** The pressure that exists inside a closed hydraulic system; the significance is that the pressure is the same everywhere within the system (assuming no velocity effects).

**I** *See* **moment of inertia**.

**$I_{DSS}$**  The highest possible drain current for a particular JFET; occurs when the gate voltage is 0 V.

**ice-water bath** A traditional way to create a known reference temperature for the cold junction of a thermocouple.

**impact pressure** The pressure in an open (pitot) tube pointed “upstream.”

**incremental optical encoder** An optical rotary encoder that has one track of equally spaced slots; position is determined by counting the number of slots that pass by a photo sensor.

**inertia** The property that explains why an object in motion will tend to stay in motion. Inertia is directly related to mass; the more mass an object has, the more energy it takes to get it moving or to stop it.

**input/output (I/O)** Data from the real world moving in and out of a computer.

**Input/Output modules** The connection points where real-world switches, relays, and the like are connected to the PLC.

**instruction set** The set of program commands that a particular microprocessor is designed to recognize and execute.

**instrument gears** Smaller gears with pitch in the 48-28 range, found in office machines and smaller industrial machines.

**instrumentation amplifier** A practical differential amplifier, usually packaged in an IC, with features such as high input resistance, low output resistance, and selectable gain.

**integral control** A feedback control strategy where the restoring force is proportional to the sum of all the past errors multiplied by time. This type of control is capable of eliminating steady-state error, but increases overshoot.

**integral horsepower (hp) motor** A motor with 1 hp or more; that is, a large motor.

**integrator** An op-amp based circuit that has an output voltage proportional to the area under the curve traced out by the input voltage.

**interfacing** The interconnection between system components.

**inverting amplifier** A simple op-amp voltage amplifier circuit with one input, where the output is out of phase with the input; one of the most common op-amp circuits.

**inverting input** The minus (–) input of an op-amp; the output will be out of phase with this input.

**I/O message and communications instructions** Allow PLCs to send and receive messages and data to other PLCs or a terminal (PC), for those applications where PLCs are networked.

**I/O** See **input/output**.

**isolation circuit** A circuit that can transfer a signal voltage without a physical electrical connection.

**iteration** One pass through the computer program being executed by the digital controller; each iteration

“reads” the set-point and sensor data and calculates the output to the actuator. See **scan**.

**JFET** See **junction FET**.

**jogging** The practice of “inching” a motor into position by repeated short bursts of power.

**junction FET** One type of an FET.

**Kelvin scale** The Kelvin temperature scale starts at absolute zero, but a Kelvin degree has the same temperature increment as a Celsius degree ( $0^{\circ}\text{C} = 273^{\circ}\text{K}$ ).

**ladder diagram** A type of wiring diagram (which resembles a ladder) used for motor-control circuits; ladder diagrams typically include switches, relays, and the devices they control.

**lag time** The time between the change of the set point and the actual movement of the controlled variable.

**LAN** See **local area network**.

**latching** See **sealing**.

**leadscrew linear actuator** A type of linear actuator that uses an electric motor to rotate a threaded shaft; the linear motion is created by a nut advancing or retreating along the threads.

**least significant bit (LSB)** The rightmost bit in a binary number, it represents the smallest quantity that can be changed—that is, the difference between two successive states.

**lever** A rigid bar pivoted at one end or somewhere in the middle and much as a pry bar used to amplify force.

**limit switch** A switch used as a proximity sensor—that is, a switch mounted so that it is activated by some moving part.

**linear actuator** A prime mover that causes movement in a straight line, such as a hydraulic cylinder.

**linearity error** A measurement error induced into the system by the sensor itself, it is the difference between the actual quantity and what the sensor reports it to be.

**linear motor** A motor that converts electric power directly into linear motion. Basically, it is a rotary brushless DC motor (BLDC) that has been rolled out flat.

**linear variable differential transformer (LVDT)** A type of linear-motion position sensor. The motion of a magnetic core, which is allowed to slide inside a transformer, is proportional to the phase and magnitude of the output voltage.

**line voltage** A single number that represents the voltage of a power system; for three-phase power, the line voltage is the voltage between any two lines.

**loading error** An error that may occur in an analog voltage signal when the circuit being driven draws too much current, thus “loading down” the voltage.

**load torque** The torque required of the motor to rotate the load.

**local area network (LAN)** A system that allows multiple units to communicate with each other, all sharing the same interconnection wire.

**lockup property** The property of high-ratio worm gears that cannot be driven backward.

**logical and shift instructions** Perform logical operations such as AND, OR, and NOT and Right and Left Shifts on words stored in memory.

**logical variable** A single data bit in those cases where a single bit is used to represent an on-off switch, motor on-off control, and so on.

**low-pass filter** A circuit that allows lower-frequency signals to pass but attenuates higher-frequency signals.

**L/R drive** A stepper motor driver circuit that uses ballast resistors in series with the motor coils to increase torque at higher stepping rates.

**LSB** *See* **least significant bit.**

**LVDT** *See* **linear variable differential transformer.**

**machine language** The set of operation codes that a CPU can execute.

**magnetic field noise** An unwanted current induced in a wire because the wire is in a time-varying magnetic field.

**magnetic motor contactor** A special-purpose relay used to switch power to a motor. The magnetic motor contactor may have large contacts to handle the motor current and may have a built-in current overload trip device.

**magnetic motor starter** *See* **magnetic motor contactor.**

**main winding** One of two windings in an AC servomotor.

**mass** The amount of material in an object; mass is related to weight in that the more mass it has, the more it will weigh.

**math instructions** PLC mathematical operations on words in memory; these operations may include addition, subtraction, multiplication, division, and others.

**membership function** Used in conjunction with fuzzy logic control, the distribution of values that have been grouped into one category.

**membrane switch** Usually a keypad with a flexible membrane over the top.

**memory** The part of the computer that stores digital data. Memory data is stored as bytes, where each byte is given an address.

**memory-mapped input/output** A system where I/O ports are treated exactly like memory locations.

**metal-oxide semiconductor FET (MOSFET)** One type of an FET.

**microcontroller** An integrated circuit that includes a microprocessor, memory, and input/output; in essence, a “computer on a chip.”

**microprocessor** A digital-integrated circuit that performs the basic operations of a computer but requires some support-integrated circuits to be functional.

**microstepping** A technique that allows a regular stepper motor to take fractional steps; it works by energizing two adjacent poles at different voltages and by balancing the rotor between.

**microswitch** A small push-button switch with a very short throw distance.

**minimum holding voltage** Minimum voltage needed to keep a relay activated.

**mnemonic** An English-like abbreviation of an operation code for PLC logical instructions.

**modem** A circuit that converts serial data from digital form into tones that can be sent through the telephone system.

**momentary-contact switch** A switch position that is spring-loaded.

**moment of inertia ( $I$ )** A mechanical property of an object that is based on its shape and mass and the axis of rotation; the larger the moment of inertia, the more torque it takes to spin the object about the designated axis.

**momentum** The property of a moving object that tends to keep it moving in the same direction.

**MOSFET** *See* **metal-oxide semiconductor FET**.

**most significant bit (MSB)** The leftmost bit in a binary number.

**Motion controllers** A control circuit specifically made to coordinate movement. A typical motion controller works with a closed loop system to coordinate the movement of two or more axes simultaneously.

**motion control** A term that refers to an electro-mechanical system wherein things move.

**MSB** *See* **most significant bit**.

**MT<sub>1</sub>** One of three terminals of a triac.

**MT<sub>2</sub>** One of three terminals of a triac.

**multiplexing** The concept of switching input signals (one at a time) through to an output; is typically used so that multiple sensors can use a single ADC (analog-to-digital converter).

**natural resonant frequency** In mechanical systems, the frequency at which a part or parts will vibrate. The resonant frequency is a function of the mass and spring constant.

**NC** *See* **numerical control**.

**negative feedback** A circuit design where a portion of the output signal is fed back and subtracted from the input signal. This results in a lower but predictable gain and other desirable properties.

**neutral** The return wire or cold side in an AC power-distribution system; the neutral wire is at ground potential.

**Newton** Unit of force in the SI system (1 N = 0.224 lb).

**node** Any unit connected to a network that can transmit and receive data.

**no-load speed** The speed of a motor when there is no external load on it; it will always be the maximum speed (for a particular voltage).

**noninverting amplifier** A simple op-amp voltage amplifier circuit with one input, where the output is in phase with the input.

**noninverting input** The positive (+) input of an op-amp; the output will be in phase with this output.

**nonvolatile memory** Computer memory such as ROM that will not lose its data when the power is turned off.

**normal force** In friction calculations, the force pushing the sliding surfaces together.

**normally closed (NC) contacts** Relay contacts that are closed in the deenergized state and open when the relay is energized.

**normally open (NO) contacts** Relay contacts that are open in the deenergized state and closed when the relay is energized.

**notch filter** A circuit that attenuates a very narrow range of frequencies.

**NPN, PNP** The two basic types of transistors, the difference being the direction of current and voltage within the device.

**null modem** A cable that allows two DTE units to communicate with each other (*see* **RS-232**).

**numerical control** A digital control system that directs machine tools, such as a lathe, to automatically machine a part.

**off-delay relay** A time-delay relay that, after being deenergized, waits a specified period of time before activating.

**on-off control** *See* **two-point control**.

**open-loop control system** A control system that does not use feedback. The controller sends a measured signal to the actuator, which specifies the desired action. This type of system is *not* self-correcting.

**open-loop gain** The gain of an amplifier when no feedback is being used; it is usually the maximum gain possible.

**operational amplifier (op-amp)** A high-gain linear amplifier packaged in an integrated circuit; the basis of many special-purpose amplifier designs.

**operation code (op-code)** A digital code word used by the microprocessor to identify a particular instruction.

**optical coupler** An isolation circuit that uses a light-emitting diode (LED) and a photocell to transfer the signal.

**optical rotary encoder** A rotary position sensor that works by rotating a slotted disk past a photo sensor.

**optical tachometer** Mounted next to a rotating shaft, a photo sensor that gives a pulse for each revolution (a stripe is painted or fixed on the shaft).

**orifice plate** A type of flow sensor whereby a restriction is placed in a pipe causing a pressure difference (between either side of the restriction) that is proportional to flow.

**overdamped** A system that has so much drag (from static or viscous friction) that its response is sluggish.

**overload device** A device that can stop the motor if an overload condition is detected. Overload is detected by sensing the temperature of the motor or the current that the motor is drawing.

**overshoot** The event that occurs when the path of the controlled variable (mechanical or electrical) approaches its destination too fast and goes beyond the set point before turning around or stopping. Underdamped systems tend to overshoot.

**parallel interface** A type of data interface where 8 bits enter or leave a unit at the same time on eight wires.

**PC** *See* **personal computer**.

**permanent magnet (PM) motors** A motor that uses permanent magnets to provide the magnetic field. The PM motor has a linear torque-speed relationship, making it desirable for control applications.

**permanent magnet (PM) stepper motor** A stepper motor that uses one or more permanent magnets for the rotor; this motor has a detent torque.

**permanent-split capacitor motor** A capacitor start motor where the start winding remains engaged during operation—that is, the motor does not have a centrifugal switch.

**personal computer (PC)** A microprocessor-based, self-contained, general-purpose computer (usually refers to an IBM or compatible computer).

**phase** As applied to motors, the number of separate field winding circuits.

**phase-control circuit** Circuit that can delay conduction for part of the AC cycle for the purpose of reducing average output voltage.

**phase margin** A form of a safety margin. When the system gain is 1, the phase margin is the amount the phase lag could be increased before the system becomes unstable.

**phase sequence** The sequence of phase voltages possible with three-phase power: ABC and ACB.

**phase voltage** The individual voltage in a multiphase power system.

**photodiode** A type of optical sensor, it increases its reverse-leakage current when exposed to light.

**photo resistor** A type of optical sensor; its resistance decreases when exposed to light.

**photo transistor** A type of optical sensor; light acts as the base current and turns on the transistor.

**photovoltaic cell** A device that converts light into electrical energy; used as an optical sensor, or as a solar cell.

**pick-and-place robot** A simple robot that does a repetitive task of picking up and placing an object somewhere else.

**PID** Stands for Proportional + Integral + Derivative; a control strategy that uses proportional, integral, and derivative feedback.

**piezoresistive effect** A property of semiconductors in which the resistance changes when subjected to a force.

**pinion** The small driver gear in a gear pass.

**PIP** Stands for Proportional + Integral + Preview; a control strategy that uses feedforward so that it knows in advance if a change is coming.

**pitch circle** If gears were solid disks, the pitch circle would be the theoretical circle that meshed gears roll on.

**pitch diameter** The diameter of the pitch circle.

**pitot tube** A velocity sensor for fluids whereby a small open tube is placed directly into the flow; the pressure in the tube is proportional to fluid velocity.

**PLC** *See programmable logic controller.*

**PLC bus** The wires in the backplane of a modular PLC system that allows the microprocessor (within the PLC) to communicate with plugged-in modules.

**plugging** A method of braking the motor by reversing the polarity of the armature windings, causing the motor to apply a reversing torque to the load.

**plug-in module** An interface circuit to the PLC, packaged as a separate module, which plugs into the PLC. Different modules support different I/O devices, from simple switch and relay output modules to ADC and DAC modules.

**PM motor** *See permanent magnet motor.*

**pneumatic control valve** Used in pneumatic systems, a valve that controls air flow to the actuators.

**pneumatic cylinder** A linear actuator powered by compressed air, it consists of a cylinder, enclosed at both ends with a piston inside. The piston is moved by admitting the compressed air into either end of the cylinder.

**pneumatic system** A system that uses compressed air to power actuators such as pneumatic cylinders.

**pneumatic time-delay relay** A time-delay relay that generates the delay by allowing air to bleed out of a spring-loaded bellows.

**polled I/O** Probably the most common method of communicating on a network, wherein a PLC asks (polls) a sensor for data only when it needs it.

**pony motor** A separate motor used to start a synchronous motor.

**port** The part of a computer where I/O data lines are connected; each port has an address.

**positive-displacement pump** A hydraulic pump that expels a fixed volume of fluid for each revolution of the pump shaft.

**positive feedback** The event that occurs when the feedback signal from the sensor is added to the set point

(instead of subtracted). Positive feedback will likely cause the system to oscillate.

**potentiometer** A variable resistor that can be used as a position sensor.

**power** A property that describes how fast energy is being used; in other words, power is energy per unit time.

**power factor** The cosine of the angle between the current and voltage. A power factor of 1 means the current and voltage are in phase, which is desirable. A lagging power factor means the current is lagging the voltage, probably due to the inductive load of motors.

**power rails** Part of a ladder diagram, the two vertical lines on each side of the diagram, which represent power lines.

**power transistor** A transistor designed to carry a large current and dissipate a large amount of heat.

**pressure-control valve** Used in a hydraulic system, an adjustable spring-loaded valve that can maintain a set pressure in a system by bleeding off excess fluid.

**pressure regulator** Used in pneumatic systems, a regulator that maintains a specified pressure.

**pressure tank** Used in pneumatic systems, a reservoir for compressed air.

**prime mover** A component that creates mechanical motion from some other energy source; the source of the first motion in the system.

**process** The physical process that is being controlled.

**process control system** The type of system where the controller is attempting to maintain the output of some process at a constant value.

**process variable (PV)** The actual value of a variable, as reported by the sensor.

**processor** Part of the PLC that executes the instructions; it is a small, microprocessor-based computer.

**program counter** A special address-holding register in a computer that holds the address of the next instruc-

tion to be executed.

**programmable gain instrumentation amplifier** An instrumentation amplifier with fixed gains that can be selected with digital inputs.

**programmable logic controller (PLC)** A small, self-contained microprocessor-based controller used primarily to replace relay logic controllers.

**programmable unijunction transistor (PUT)** A solid-state device that performs the same function as a UJT except that the trigger voltage is adjustable.

**program file** Contains the program that the PLC executes. The program file is part of the PLC memory (the other part being data files).

**program memory** When applied to PLCs, the section of memory (RAM or EEPROM) where the PLC program is stored.

**programming port** A connector on a PLC through which the program is downloaded.

**proportional band** A term used to describe the gain of a proportional control system. It is the change in error that causes the controller output to change its full swing.

**proportional control** A feedback control strategy where the controller provides a restoring force that is proportional to the amount of error.

**proximity sensor** A sensor that detects the physical presence of an object.

**psi** Pounds per square inch, a unit of pressure.

**pull-in current** Minimum current needed to activate a relay.

**pull-in voltage** Minimum voltage needed to activate a relay.

**pull-out torque** The maximum torque that an AC motor can provide just before it stalls. However, unlike the DC motor, this maximum torque condition occurs at about 75% of the unloaded speed.

**pulse-width modulation (PWM)** A method of controlling an electric motor's speed by providing pulses that are at a constant DC voltage. The width of the pulses is varied to control the speed.

**push-button switch** A momentary switch activated by pushing.

**PUT** See **programmable unijunction transistor**.

**PV** See **process variable**.

**PWM** See **pulse-width modulation**.

**RAM** See **random-access memory**.

**random-access memory (RAM)** Sometimes called read/ write memory, a memory arrangement using addresses where data can be written in or read out; RAM loses its contents when the power is turned off.

**Rankine scale** The Rankine temperature scale starts at absolute zero, but a Rankine degree has the same temperature increment as a Fahrenheit degree ( $0^{\circ}\text{F} = 460^{\circ}\text{R}$ ).

**rated speed** The speed of a motor when producing its rated horsepower.

**reaction-curve method** A method of tuning a PID controller that involves opening the feedback loop, manually injecting a step function, and measuring the system response. The PID constants are computed based on these data.

**read-only memory (ROM)** Similar to RAM in that it is addressable memory, but it comes preprogrammed and cannot be written into; also, it does not lose its data when the power is turned off.

**read/write (R/W) line** A control signal that goes from the microprocessor to memory.

**real time** Refers to a computer that is processing data *at the same time* that the data are generated by the system.

**reciprocating piston compressor** An air compressor that uses pistons and one-way valves to compress the air.

**reduced voltage-starting circuit** A special circuit that limits the motor current during start-up.

**reed relay** A small relay with contacts sealed in a tube and activated by a magnetic field.

**regulator system** A control system that maintains an output at a constant value.

**resistance temperature detector (RTD)** A temperature sensor based on the fact that the resistance of a metal wire will increase when the temperature rises.

**resolution** In digital-to-analog conversion, the error that occurs because digital data can only have certain discrete values. The smallest increment of data that can be detected or reported. For a digital system, the resolution is usually the value of the least significant bit.

**return wire** See **neutral**.

**rise time** The time it takes for the controlled variable to rise from 10 to 90% (of its final value).

**robot** A servomechanism control system in the form of a machine with a movable arm.

**ROM** See **read-only memory**.

**rotary switch** A rotating knob that activates different switch contacts.

**rotating field** The phenomenon of what is apparently happening in the stator coils of an AC motor. Even though the coils themselves are stationary, the magnetic field is transferred from coil to coil sequentially, producing the effect that the field is rotating.

**rotor** A basic part of an AC motor, the part in the center that actually rotates. If the field poles are stationary (as they usually are), then the rotor is known as the armature.

**RS-232 standard** A serial data transmission standard that specifies voltage levels and signal protocol between a DTE (computer) and a DCE (modem).

**RTD** See **resistance temperature detector**.



**run mode** The operational mode of the PLC when it is actually executing the control program and hence performing some control operation.

**run windings** The main stator windings in a single-phase motor.

**R/W** See **read/write line**.

**sample rate** The times per second (or per minute) that a sensor is read; for a digital controller, the sample rate usually corresponds to the scan rate.

**sample-and-hold circuit** A circuit that can temporarily store or remember an analog voltage level.

**scan** The event when the processor makes one pass through all instructions in the control program loop. See **iteration**.

**SCR** See **silicon-controlled rectifier**.

**sealed current** Current required to keep a relay energized.

**sealing** A relay wired such that once energized its own contacts take over the job of providing power to the coil; it will stay energized, even if the original energizing signal goes away.

**Seebeck effect** The property used by a thermocouple, a voltage proportional to temperature developed in a circuit consisting of junctions of dissimilar metal wires.

**sensor** Part of the control system that monitors the system output (such as temperature, pressure, or position), the sensor converts the output movement of the system into an electric signal, which is fed back to the controller.

**sensorless drive** A class of AC-motor-control units that does not use a sensor. A type of vector drive. See **vector drives**.

**sequentially controlled system** A control system that performs a series of actions in sequence, an example being a washing machine.

**serial interface** A type of interface where data are transferred 1 bit after the other on a single wire.

**series-wound motor** A motor that has the field windings connected in series with the armature; the series-wound motor has a high starting torque and a high no-load speed.

**servomechanism** An electromechanical feedback control system where the output is linear or rotational movement of a mechanical part, such as a robot.

**set point (SP)** An input into the controller, it represents the desired value of the controlled variable.

**settling time** The time it takes for the system response to settle down to some steady-state value.

**shaded pole** A small metal ring around the end of the electromagnetic pole of an AC relay, for the purpose of keeping the relay from “buzzing” at 60 Hz.

**shaded-pole motor** A single-phase induction motor, usually small, that does not use a start winding. Instead, a modification to the single-phase poles causes the motor to have a small starting torque.

**shunt-wound motor** A motor that has the field windings connected in parallel with the armature windings; the shunt-wound motor has a measure of natural speed regulation—that is, it tends to maintain a certain speed despite load changes.

**signal common** The common voltage reference point in a circuit, usually the negative terminal of the power supply (or battery).

**signal return** See **signal common**.

**silicon-controlled rectifier (SCR)** A bistable, three-terminal, semiconductor device used to switch power to a load.

**single-acting cylinder** A spring-loaded pneumatic cylinder; air pressure drives the piston in one direction, and spring pressure returns it.

**single-board computer** A premade microprocessor-based computer assembled onto a single printed-circuit card.

**single-ended voltage** A signal voltage that is referenced to the ground.

**single-phase AC** The type of AC supplied to residences (and industry). It consists of a single alternating waveform that makes 60 complete cycles per second.

**single-point ground** A single connection point in a system, where all signal commons are connected together and then connected to an earth ground.

**single-pole/double-throw switch (SPDT)** A switch contact configuration.

**single-pole/single-throw switch (SPST)** A switch contact configuration.

**single-step mode** Operating the stepper motor at a slow enough rate so that it can be stopped after any step without overshooting.

**slew mode** Stepping the stepper motor at a faster rate than the single-step mode; used to move to a new position quickly. The motor will overshoot if the speed is not ramped up or down slowly.

**slide switch** Similar to a toggle switch except the handle slides back-and-forth.

**slider (wiper)** The moving contact in a potentiometer, usually the center of three terminals.

**sliding friction** The frictional drag force on two dry sliding objects.

**slip** The difference between the rotor speed of an induction motor and the synchronous speed of the stator field; the amount of slip is typically less than 10%.

**slip rings** Sliding electrical contacts that allow power to be fed to the rotor of the synchronous motor.

**slotted coupler** An optical proximity sensor that is activated when an object “cuts” a light beam.

**snubber** A circuit that prevents a fast voltage rise across an SCR, for the purpose of keeping it from false firing.

**solenoid** An electromechanical device that uses an electromagnet to produce short-stroke linear motion.

**solid-state relay (SSR)** A solid-state switching device used as a relay.

**solid-state time-delay relay** A time-delay relay that generates the delay with an electronic circuit.

**source** One of three terminals in an FET.

**SP** *See* **set point**.

**SPDT** *See* **single-pole/double-throw switch**.

**speed regulation** In general, a motor’s ability to maintain its speed under different loads; specifically, a percentage based on no-load speed and full-load speed.

**split-phase** A general term applied to an induction motor that has two sets of windings but operates on single-phase AC with some provision to phase-shift the AC for the second set of windings.

**split-phase control motor** A two-phase motor, usually small, powered by single-phase AC using a capacitor for the phase shift. The direction is reversible, making these motors useful in controlling back-and-forth motion.

**spool** The moving part within the bidirectional control valve.

**SPST** *See* **single-pole/single-throw switch**.

**spur gear** A type of circular gear with radial teeth (the most common type of gear).

**squirrel cage rotor** The type of rotor used in an induction motor—named as such because someone thought it looked like a squirrel cage.

**SSR** *See* **solid-state relay**.

**stall** When the load on the AC motor is increased (about 75% of unloaded speed), the motor torque drops back abruptly, and the motor stalls and stops.

**stalling** A situation wherein the motor cannot rotate because the load torque is too great.

**stall torque** The torque of the motor when the shaft is prevented from rotating; for DC motors it will always be the maximum torque (for a particular voltage).

**start windings** A second set of windings (besides the run windings) in the single-phase induction motor; the start windings are usually only engaged during the starting process.

**starting torque** The motor torque available when the motor is first started; the starting torque of most AC motors is less than its maximum torque but more than its rated load.

**static friction** The friction force that must be overcome to get an object at rest to move; for a particular object, static friction is greater than sliding friction.

**static pressure** The pressure measured when the open tube is directed perpendicularly to the flow.

**stator** A basic part of an AC or stepper motor, a stationary collection of coils that surrounds the rotor and consists of field poles (electromagnets).

**steady-state error** The error that remains after the transient response has died away.

**step change** The event that occurs when a discrete change is made to the set point.

**stepper motor** A motor that rotates in steps of a fixed number of degrees each time it is activated.

**strain** The deformation (per unit length) as a result of stress.

**stress** Subjecting an object to tension or compression forces. Stress is the force per unit area within the object.

**stroke** The overall linear travel distance of a linear actuator.

**summing amplifier** An op-amp circuit that has multiple inputs and one output. The value of the output voltage is the sum of the individual input voltages.

**swash plate** Part of the axial piston pump; when the angle of the swash plate is changed, the flow from the pump is changed.

**switch wafer** Part of a rotary switch.

**synchronous condenser** A term applied to a synchronous motor that is being used solely for power factor correction.

**synchronous motor** An AC motor that runs at the synchronous speed, which means an exact multiple of the line frequency. A synchronous motor has no slip.

**synchronous speed** The speed of the rotating field in the stator, which is always an exact multiple of the line frequency. A synchronous motor spins at the synchronous speed, whereas an induction motor turns somewhat slower than the synchronous speed.

**TF** *See* **transfer function**.

**thermal time-delay relay** A time-delay relay that generates the delay by allowing a bimetallic strip to heat and bend, thus activating a switch.

**thermistor** A temperature sensor based on the fact the resistance of some semiconductors will decrease as the temperature increases.

**thermocouple** A temperature-measuring sensor made from the junction of two dissimilar metal wires; when the junction is heated, a small voltage is generated.

**three-phase AC** The type of AC available to industry. It consists of three AC waveforms (called phases) on three wires, where each phase is delayed 120° from the previous phase and all phases operate at 60 Hz.

**three-phase stepper motor** A motor with three separate sets of field coils; usually found with VR motors.

**three-position control** Similar to the two-point system except the actuator can exert force in two directions, so the system is either off-up-down, off-right-left, and so on.

**three-position switch** Switch with a center position.

**threshold detector** A circuit that provides a definite off-on signal when an analog voltage rises above a certain level.

**thumbwheel switch** Switch that rotates a drum to select numeric data.

**thyristor** A class of four-layer semiconductor devices (such as PNP) that are inherently bistable; the SCR and triac are thyristors.

**time-delay loop** A programming technique where the computer is given a “do-nothing” job such as counting to some large number for the purpose of delaying time.

**time-driven operation** In a sequentially controlled system, an action that is allowed to happen for a specified period of time. This is an example of open-loop control.

**timer relay (TR)** A relay with a built-in time delay mechanism so that the contacts close or open a specified time after the relay is energized.

**toggle switch** A manually operated device that connects or disconnects power.

**toothed-rotor tachometer** A tachometer that consists of two parts: a toothed, iron-based wheel attached to the rotating member, and a stationary sensor. The sensor gives a pulse each time a tooth passes by.

**torque** In general, the measure of the motor’s strength in providing a twisting force; specifically, the product of a tangential force times the radius. Torque is used in rotational systems just as force is used in linear systems.

**torque-speed curve** A graph of a motor’s torque versus speed; can be used to predict the motor’s speed under various load conditions.

**TR** See **timer relay**.

**transconductance ( $g_m$ )** The gain of an FET, which is the change in drain current divided by the change in gate voltage.

**transducer** A term used interchangeably with *sensor*. Literally means that energy is converted, which is what a sensor does.

**transfer function** A mathematical relationship between the input and output of a control system component:  $TF = \text{output}/\text{input}$ . (In this text, *transfer function* and *gain* are used interchangeably, although this is a simplification.)

**transient response** The actual path that the controlled variable takes when it goes from one position to another.

**triac** A bistable, three-terminal, solid-state device that switches power.

**trigger circuit** A circuit used to generate the turn-on pulse for an SCR or a triac.

**tuning** When applied to a control system, this term refers to the process of making adjustments to the gain constants  $K_P$ ,  $K_I$ ,  $K_D$  until system performance is satisfactory.

**turbine** A flow sensor based on having the fluid rotate a propeller of some kind.

**two-capacitor motor** A capacitor start motor that uses one value of capacitance for starting and switches to a lower value of capacitance for running.

**two-phase stepper motor** A motor with two field circuits. This motor requires polarity reversals to operate and hence is bipolar.

**two-point control** A simple type of control system where the actuator is either on full force or off; also called *on-off control*.

**two-position switch** A toggle or slide switch with two positions.

**UART** See **universal asynchronous receiver transmitter**.

**UJT** See **unijunction transistor**.

**underdamped** A system that has relatively little damping so that it responds quickly and tends to overshoot.

**unijunction transistor (UJT)** A bistable, three-terminal solid-state device that primarily triggers an SCR.

**unipolar motor** A motor that does not require polarity reversals. A four-phase stepper motor is unipolar.

**universal asynchronous receiver transmitter (UART)** A special purpose integrated circuit that converts data from parallel to serial format and vice versa.

**universal motor** An electric motor that can run on AC or DC power; it is essentially a series DC motor.

**vane motor** A device that creates rotary motion from pneumatic or hydraulic pressure. The construction is similar to a vane pump.

**vane pump** A hydraulic pump that uses an offset rotor with retractable vanes.

**variable-displacement pump** A hydraulic pump that can be adjusted to change the flow while keeping the speed of rotation constant.

**variable-frequency drive (known as volts-per-hertz (V/Hz) drive)** A type of AC-motor-control unit that controls speed by changing the motor's frequency and voltage using solid-state control units.

**variable reluctance (VR) stepper motor** A motor that uses a toothed iron wheel for the rotor and consequently can take smaller steps but has no detent torque.

**variable-reluctance sensor** A magnet with a coil of wire around it. It can detect the movement of iron in the vicinity. Typically used with the toothed-rotor tachometer.

**vector drives** A class of AC-motor-control units that are based on the principle that the motor current is the vector sum of the flux-producing current and the speed/torque-producing current. Some vector drives can control an AC motor down to 0 rpm.

**venturi** A type of flow sensor whereby fluid is forced into a smaller channel, which increases its velocity; the higher velocity fluid has a lower pressure (than the fluid in the main channel), and the pressure difference is proportional to velocity.

**$V_{GS(off)}$**  The gate voltage necessary to turn the drain current off (for a JFET).

**virtual ground** A point in a circuit that will always be practically at ground voltage but that is not physically connected to the ground.

**viscous friction** A drag force experienced when a lubricant is used between the objects so that the objects do not actually touch; the drag force, which is proportional to velocity, comes from the layers of lubricant slipping over each other.

**vision sensor** A system whereby a computer analyzes a video image for the purpose of inspecting parts or guiding a robot or machine.

**volatile memory** Computer memory such as RAM that will lose its data when the power is turned off.

**voltage follower** A very simple but useful op-amp circuit with a voltage gain of 1, used to isolate circuit stages and for current gain.

**volts-per-hertz (V/Hz) drive** *See variable-frequency drive.*

**weight** Technically, a downward force exerted by a mass, caused by gravity.

**window comparator** A comparator with built-in hysteresis, that is, two thresholds: an upper switch-in point and a lower switch-in point.

**windup** Associated with integral feedback, the problem that occurs when too much error accumulates as a result of the system saturating before it can achieve the state that it theoretically should.

**wiper** The center and/or moveable contact in a switch. *See slider.*

**wire-wound potentiometer** A variable resistor that uses a coil of resistance wire for the resistive element; can be used as a position sensor.

**word** A unit of digital data that a particular computer uses; common word sizes are 4, 8, 16, and 32 bits.

**worm** The spiral-looking gear in a worm gearbox.

**worm gear** The circular gear that the worm meshes with in a worm gearbox.

**wye-connection** One of two ways in which to connect three-phase motor or generator coils; with the wye-connection, each of the three coils is connected between a phase wire and neutral.

**Young's modulus** A constant that relates stress and strain for a particular material (Young's modulus = stress/strain).

**zero-voltage switching** A technique of switching on an AC device (such as an SCR) only at the beginning of the cycle when the voltage is zero anyway. This system eliminates sharp voltage rises at turn-on and reduces the generation of high-frequency interference.

# APPENDIX D

## Answers to Odd-Numbered Exercises

### Section 1.1

1.1

- a. Set point → Controller → Actuator → Process  
→ Controlled variable
- b. Set point specifies the desired system output. The controller generates a drive signal to the actuator. The actuator moves the process. There is no feedback in an open-loop control system, so the controller cannot correct any errors that may occur “downstream.”
- c. The operating characteristics of the components must be well-known and stable or predictable.
- d. Simplicity, less hardware

1.3

The error is initially  $15^\circ$  ( $45^\circ$ – $30^\circ$ ). This error signal causes the controller to drive the robot arm toward the  $30^\circ$  position. When the arm reaches  $30^\circ$ , there is no more error, so the controller directs the arm to stop moving.

1.5

$$V_{\text{pot}} = \frac{.1 \text{ V}}{\text{deg}} \times 45 \text{ deg} = 4.5 \text{ V}$$

1.7

Two sets of input/output data are given; either can be used.

$$\text{TF}_{\text{motor}} = \frac{500 \text{ rpm}}{6 \text{ V}} = 83.3 \text{ rpm/V}$$

### Section 1.2

1.9

#### Analog control system

Uses analog circuits such as op-amps, which are hard-wired to do the job.

No appreciable delay time from input to output.

#### Digital control system

Uses a microprocessor or microcontroller under program control.

Works on the basis of a program loop, which is a 3-step process:

1. Inputs read
2. Outputs calculated
3. Outputs sent out

Usually requires using ADCs or DACs.

### Section 1.3

1.11

A **process control system** is usually trying to keep some parameters constant in a continuous process, such as the thickness of a steel plate from a rolling mill, or it is controlling a sequence of events, such as filling a tank, mixing, and draining.

A **servomechanism** usually is a position-control system, which controls the movement of some device, such as a robot arm.

## 1.13

- a. Time driven—defrost cycle on a “frost-free” refrigerator
- b. Event driven—Gasoline filler nozzle automatically shuts off when tank is full
- c. Time and event driven—A batch mixing process whereby (1) the fill valve opens until vat is full; (2) mixes for 2 minutes

## Section 2.1

## 2.1

**ALU** (Arithmetic Logic Unit)—performs calculations.

**Control Unit**—manages the data flow within the computer, such as reading and executing program instructions.

**CPU** (Central Processing Unit)—The “heart” of a computer, includes the ALU and control unit. A microprocessor is a CPU.

**Memory**—Consists of memory cells organized in groups of 8 bits (byte). Each byte is given a unique address. Data are stored and retrieved on the basis of their address. The memory is used to store the program, as well as the program results.

**I/O** (Input/Output)—The input port is a channel through which data from the “outside world” enter the computer. The output port is a channel through which data leave the computer.

## 2.3

**Address bus**—A group of wires that transports the memory or I/O address from the microprocessor.

**Data bus**—A group of wires that transports the data to and from the CPU, from memory and I/O devices.

**Control bus**—A group of wires that transports timing and control signals from the CPU to memory and I/O.

## 2.5

Decimal value = 218

## Section 2.2

## 2.7

Final value in accumulator = 4

## Section 2.3

## 2.9

**Parallel data port**—Digital data is transferred from the computer to the “outside world” in 8- (or 16) bit chunks on 8 (or 16) wires.

**Serial data port**—Digital data are transferred on a single wire bit by bit (one bit at a time).

## 2.11

$$V_{\text{out}} = \frac{204 \times 9 \text{ V}}{256} = 7.17 \text{ V}$$

## 2.13

$$\begin{aligned} \text{Output} &= \frac{3.7 \text{ V} \times 255}{12 \text{ V}} = 78.6 \approx 79 \\ &= 1001111_{\text{Binary}} \end{aligned}$$

## Section 2.4

## 2.15

“Real time” computing means that the computer is connected directly to the “real world” system and is performing calculations on the system data as it comes in. Real time computing is necessary for control systems because the controller response must be calculated immediately upon receipt of sensor data.

## 2.17

A computer generates a time delay by using a “time-delay loop,” which is a section of the program that counts up to a large number: the larger the number, the longer the delay.

## Section 2.5

## 2.19

A **microcontroller** is a “computer on a chip.” It contains a **microprocessor**, memory, I/O ports, and perhaps other features. Newer microprocessors are faster and have a more sophisticated set of instructions than a microcontroller, but they require external components, such as memory, to work.



## 2.21

1. An expansion card(s) for the PC with at least two input ports (for the two position sensors), and two output ports (for the two motors). Each input port requires an ADC and each output port requires a DAC (the ADCs and DACs may be already on the expansion card).
2. Two power amplifiers for the motors, to take the signal from the DACs and to boost the power to drive the motors
3. Software to perform the control operation. The software would have to read in the sensor data from the input expansion card, and send the motor drive signals to the output expansion card.

## Section 3.1

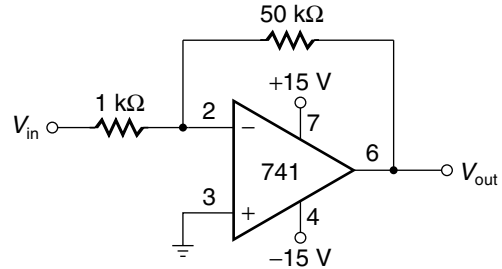
## 3.1

- a.  $10(2 - 1) = 10 \text{ V}$
- b.  $10(-3 - -2.5) = -5 \text{ V}$
- c.  $10(2.5 - -3) = 55 \text{ V}$
- d.  $10(-2.5 - -4) = 15 \text{ V}$
- e.  $10(-1.5 - 1) = -25 \text{ V}$
- f.  $10(1.5 - 3) = -15 \text{ V}$

## 3.3

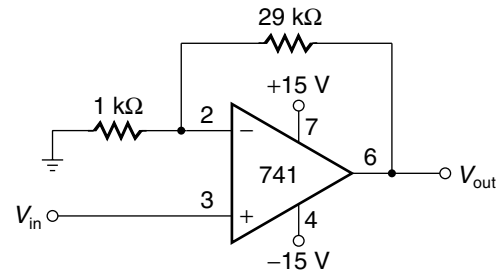
The input goes directly into the + input of the op amp; therefore, the output will be in phase with the input, and the input resistance is very high. The output is connected back to the - input, and because the two op amp inputs are virtually the same voltage, the output voltage will be the same as the input voltage, so the gain is 1.

## 3.5



Select  $R_i = 1 \text{ k}\Omega$ , then  $R_f = -A R_i = -50 \times 1 \text{ k}\Omega = 50 \text{ k}\Omega$

## 3.7

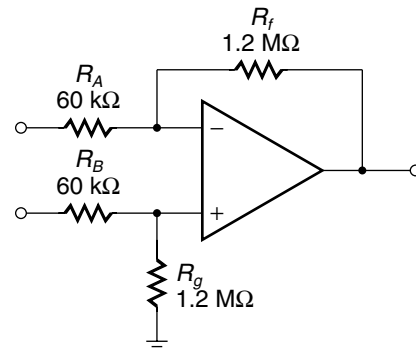


Select  $R_i = 1 \text{ k}\Omega$

$$A_v = 30 = \frac{R_f}{1 \text{ k}\Omega} + 1$$

$$R_f = 29 \times 1 \text{ k}\Omega = 29 \text{ k}\Omega$$

## 3.9

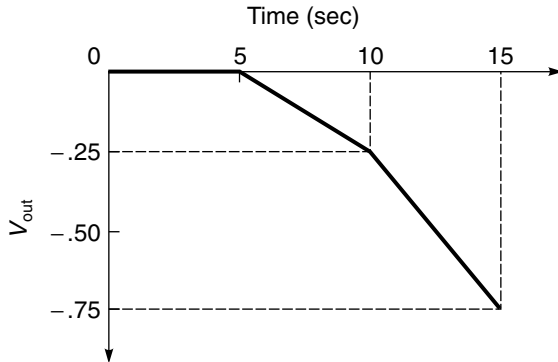
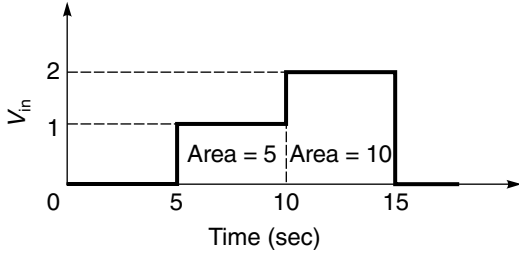


Select  $R_A$  and  $R_B$  to be ten times higher than source impedance:

$$R_A = R_B = 10 R_{\text{source}} = 60 \text{ k}\Omega$$

$$R_f = R_g = 20 \times 60 \text{ k}\Omega = 1.2 \text{ M}\Omega$$

3.11

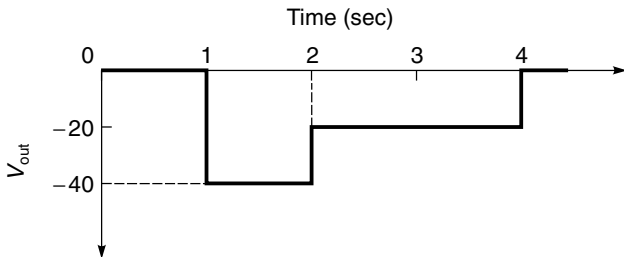
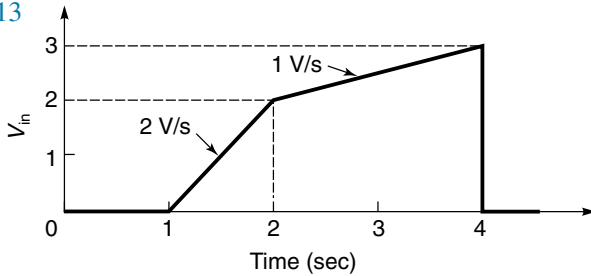


$$V_{out}(5\text{ s}) = -0.05 \times 0 = 0\text{ V}$$

$$V_{out}(10\text{ s}) = -0.05 \times 5 = -0.25\text{ V}$$

$$V_{out}(15\text{ s}) = -0.05(5 + 10) = -0.75\text{ V}$$

3.13



$$V_{out(0-1\text{ s})} = -20 \times 0\text{ V/s} = 0\text{ V}$$

$$V_{out(1-2\text{ s})} = -20 \times 2\text{ V/s} = -40\text{ V}$$

$$V_{out(2-4\text{ s})} = -20 \times 1\text{ V/s} = -20\text{ V}$$

$$V_{out(4s+)} = -20 \times 0\text{ V/s} = 0\text{ V}$$

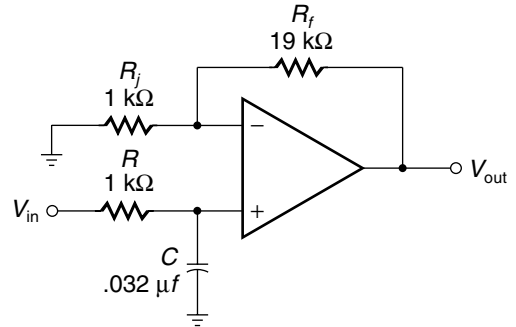
3.15

- a.  $A_{Vdb} = 20 \log 3300 = 70.4\text{db}$
- b.  $3300 \approx 4000 = 2 \times 2 \times 10 \times 10 \times 10$   
substituting  $6 + 6 + 20 + 20 + 20 = 72\text{db}$

3.17

- a.  $A_v = 10^{(50\text{db}/20)} = 10^{2.5} = 361$
- b.  $50\text{db} \approx 52\text{db} = 6 + 6 + 20 + 20$   
substituting  $2 \times 2 \times 10 \times 10 = 400$

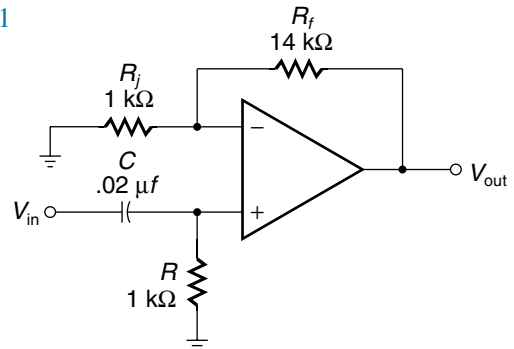
3.19



$$C = \frac{1}{2\pi(5 \times 10^3)(1 \times 10^3)} = 0.032\ \mu\text{f}$$

Select  $R_i = 1\text{ k}\Omega$   
 $R_f = 19 \times 1\text{ k}\Omega = 19\text{ k}\Omega$

3.21



$$C = \frac{1}{2\pi(8 \times 10^3)(1 \times 10^3)} = 0.02 \mu\text{f}$$

Select  $R_f = 1 \text{ k}\Omega$

$$R_f = 14 \times 1 \text{ k}\Omega = 14 \text{ k}\Omega$$

for Gain = 1,  $R_f = R_A$     Select 10 k $\Omega$

3.27

$$C = \frac{t}{5 R_s} = \frac{0.2\text{s}}{5 \times 1\text{k}\Omega} = 40 \mu\text{f}$$

### Section 3.3

3.29

A **ground loop** is a current which can exist in a wire that is connected to the earth at two different places. Large currents can result, because even a few volts' difference across a low resistance ground wire develops a large current.

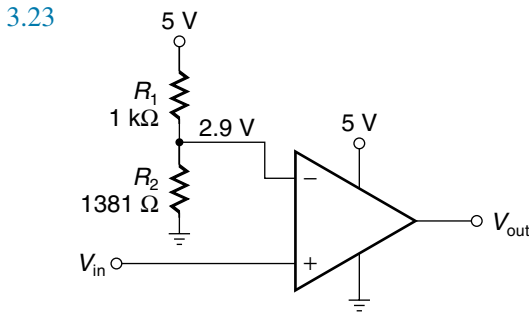
3.31

$$\text{Total wire resistance} = 300\text{ft} \times \frac{.01 \Omega}{\text{ft}} = 3 \Omega$$

$$\text{ground loop current } I = \frac{V}{R} = \frac{5\text{V}}{3 \Omega} = 1.67 \text{ A}$$

3.33

When  $V_{in}$  is pulled to  $\approx 0 \text{ V}$  by the TTL logic circuit going "low," a voltage is developed across the LED. Within the unit, light from the LED strikes the base of a photo transistor and turns it on. The transistor conducts and pulls  $V_{out}$  to  $\approx 0 \text{ V}$ . A pull-up resistor on  $V_{out}$  pulls it up to 5 V when the transistor is off.



Select  $R_1 = 1 \text{ k}\Omega$

Solve for  $R_2$ .  $R_2 = 1381 \Omega$

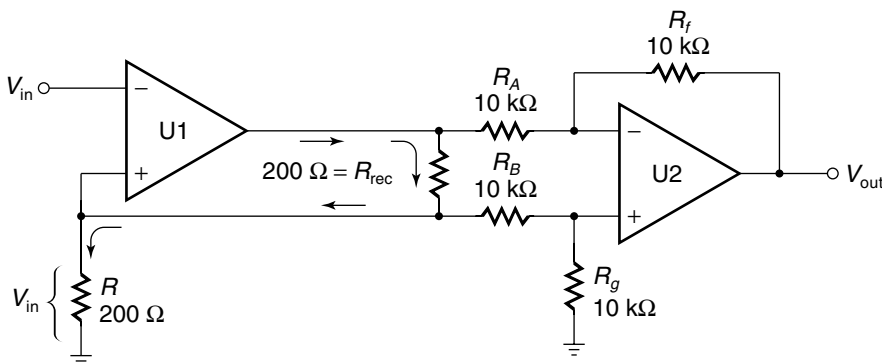
### Section 3.2

3.25

(See figure at bottom of page)

$$R = \frac{V}{I} = \frac{1 \text{ V}}{5 \text{ mA}} = 200 \Omega$$

If we select the receiver resistor ( $R_{rec}$ ) to be 200  $\Omega$ , the gain of the differential amp should be 1.



## 3.35

The principle of **magnetic shielding** is that the shield material (like steel) is so conducive to magnetic fields, that the field would rather go in the shield, leaving the adjacent area relatively free of magnetic fields.

## 3.37

A **single-point ground** is where all the shields and grounds are connected together at one place, and connected to the earth. It is important because it prevents “ground loops.”

## 3.39

Wire resistance for #22 wire (from Table 3.2) =  $16.14\Omega/1000\text{ft}$

Total wire length =  $500\text{ft} + 500\text{ft} = 1000\text{ft}$

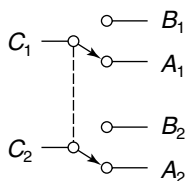
Total wire resistance =  $1000\text{ft} \times \frac{16.14\Omega}{1000\text{ft}} = 16.14\Omega$

Voltage drop  $V = I \times R = 50\text{mA} \times 16.14\Omega = .8\text{V}$

## Section 4.1

## 4.1

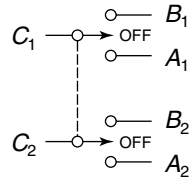
DPDT toggle switch:



Wipers  $C_1$  and  $C_2$  move together: when up, they make contact with  $B_1$  and  $B_2$ ; when down, they make contact with  $A_1$  and  $A_2$ .

## 4.3

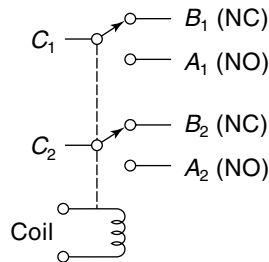
The 2SHX191 switch is a two-pole on-off-on momentary switch. The rest position is off in the middle. It may be activated up or down under spring pressure.



## Section 4.5

## 4.5

DPDT relay:



When the coil is energized, contacts  $C_1$  and  $C_2$  are pulled down.

## 4.7

For the Y4-V52 relay:

coil voltage =  $6\text{ Vdc}$  (from table)

$$\text{coil current} = \frac{V}{R} = \frac{6\text{ V}}{52\ \Omega} = 115\text{ mA}$$

contact voltage  $125\text{ Vac}$  or  $28\text{ Vdc}$

contact current  $2\text{ A typ}$  ( $3\text{ A max}$ )

## 4.9

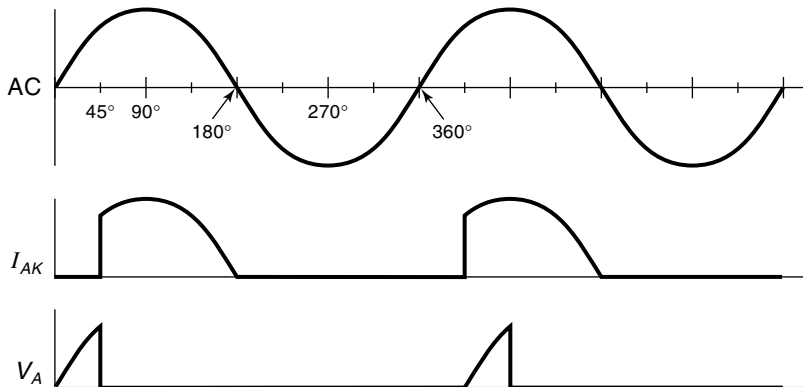
All relays in table meet the  $1.5\text{ A}$  requirement. Auto relay should use  $12\text{ Vdc}$  coil.

Best choice: Y2 – V185

## Section 4.3

## 4.11

$$I_B = \frac{I_C}{h_{FE}} = \frac{5\text{ A}}{60} = 83\text{ mA}$$



4.13

$$I_C = h_{FE} I_B = 40 \times 25 \text{ mA} = 1 \text{ A}$$

$$\text{Approximate: } I_E = I_C = 1 \text{ A}$$

$$\text{Exact: } I_E = I_C + I_B = 1 \text{ A} + 25 \text{ mA} = 1.025 \text{ A}$$

4.15

$$V_B = V_{\text{tot}} \frac{R_2}{R_1 + R_2} = 10 \text{ V} \times \frac{100 \Omega}{1.3\text{k} + 100} = 0.714 \text{ V}$$

Using input curves,

$$I_B \approx 1.1 \text{ mA. } I_C = h_{FE} I_B = 70 \times 1.1 \text{ mA} = 77 \text{ mA}$$

4.17

Transistor operating modes:

**Class A**—Transistor is biased so that it is approximately half on, with no signal applied. This way, the output can swing either up or down (from the midpoint).

**Class B**—Transistor is biased so that it is at the brink of turning on, but it is off if no signal is applied. This way, any positive-going input voltage will begin to turn the transistor on.

**Class C**—Transistor is biased so it is completely off, and the input voltage, when applied, will turn the transistor completely on. So the transistor acts like a switch, either completely on, or completely off.

4.19

$$\text{Select MJE18204 } I_C = 5 \text{ A} \quad h_{FE(\text{min})} = 18 \\ P_p = 75 \text{ W}$$

4.21

**JFET**

1. Gate voltages must be negative.
2. Input resistance is very high.

**MOSFET**

1. Gate voltages can be positive.
2. Input resistance is almost infinite because gate is capacitively coupled.

4.23

For  $V_{GS(\text{th})} = 2 \text{ V}$ :

$$V_{GS(\text{active})} = 5 \text{ V} - 2 \text{ V} = 3 \text{ V}$$

$$I_D = V_{GS} g_{fs} = 3 \text{ V} \times 12 \text{ mho} = 36 \text{ A (max) which exceeds the limits of 24 A}$$

For

$$V_{GS(\text{th})} = 4 \text{ V:}$$

$$V_{GS(\text{active})} = 5 \text{ V} - 4 \text{ V} = 1 \text{ V}$$

$$I_D = V_{GS} g_{fs} = 1 \text{ V} \times 12 \text{ mho} = 12 \text{ A (min)}$$

## Section 4.4

4.25

Turning an SCR off:

1. Open the load circuit with another device, like a switch.
2. Momentarily short the SCR with another device.
3. Use a second SCR, capacitively coupled to the primary SCR (pulls down anode voltage)

4.27

(See figure at top of previous page)

Waveforms for SCR, which are triggered at  $45^\circ$ .

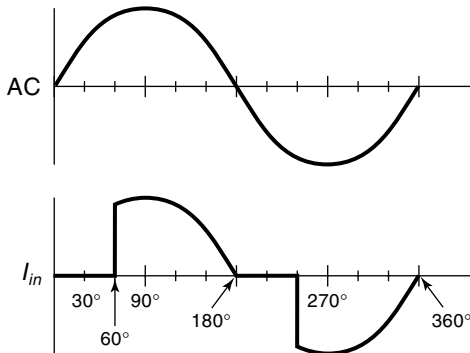
4.29

Select 2N6397. Load current = 12 A, Voltage limit = 400 V

## Section 4.5

4.31

Load current ( $I_m$ ) waveform for Triac triggered at  $60^\circ$ .



4.33

Delay time =  $8.33 \text{ ms} - 2 \text{ ms} = 6.33 \text{ ms}$

4.35

Conduction time =  $8.33 \text{ ms} - 3.52 \text{ ms} = 4.81 \text{ ms}$

## Section 4.6

4.37

The purpose of  $R_1$  is to limit the current that is charging capacitor  $C$ . By controlling the resistance of  $R_1$ , the triggering point of the UJT is controlled.

4.39

When the voltage between  $R$  and  $C$  get high enough, the diac switches into its conduction mode, which in turn triggers the triac into conduction.

## Section 5.1

5.1

**Static friction**—“Start-up” friction; the force it takes to get something moving from a rest state

**Sliding friction**—“Running” friction; the continuous drag on an object once it’s moving. Sliding friction is less than static friction.

**Viscous friction**—The drag on an object running with a lubricant. Viscous friction is proportional to velocity.

5.3

a. Force to get table started moving:

$$F = \mu N = .62 \times 40 \text{ lb} = 24.8 \text{ lb}$$

b. Force to keep table moving:

$$F = \mu N = .48 \times 40 = 19.2 \text{ lb}$$

5.5

$$k = \frac{F}{\text{deflection}} = \frac{1 \text{ oz}}{0.25 \text{ in.}} = 4 \text{ oz/in.}$$

5.7

$$F = ma = \frac{0.313 \text{ lb} \cdot \text{s}^2}{\text{ft}} \times \frac{2 \text{ ft}}{\text{s}^2} = 0.63 \text{ lb}$$

5.9

$$F = ma = 5 \text{ kg} \times \frac{0.5 \text{ m}}{\text{s}^2} = 2.5 \text{ N}$$

5.11

The heavier part will take more power to accelerate, because it has a larger moment of inertia ( $I$ ).

5.13

$$\alpha = \frac{T}{I} = 1.33 \text{ radians/s}^2$$

$$t = \frac{w}{\alpha} = 7.87 \text{ s}$$

5.15

$$\alpha = \frac{T}{I} = 0.152 \text{ rad/s}^2$$

$$t = \frac{w}{\alpha} = 68.9 \text{ s}$$

5.17

Force from a lever applied to cap

$$d_1 = 18 \text{ in} - 3 \text{ in} = 15 \text{ in}, d_2 = 3 \text{ in}$$

$$F_2 = \frac{10 \text{ lb} \times 15}{3 \text{ in}} = 50 \text{ lb}$$

$$\text{Stroke at cap } s_2 = \frac{2 \text{ in} \times 3 \text{ in}}{15 \text{ in}} = .4 \text{ in}$$

## Section 5.2

5.19

$$\begin{aligned} \text{Power} &= VI = 220 \text{ V} \times 15 \text{ A} = 3300 \text{ w} \\ &= 11281 \text{ Btu/hr} \end{aligned}$$

5.21

$$I = \frac{P}{V} = \frac{26.6 \text{ w}}{120 \text{ V}} = 0.22 \text{ A}$$

5.23

$$I = \frac{P}{V} = \frac{28.8 \text{ w}}{120 \text{ V}} = 0.24 \text{ A}$$

## Section 5.3

5.25

A car that keeps on bouncing would be **underdamped**.

A car body that dipped just once after going over the bump would be **critically damped**.

A car with very stiff shock absorbers would not even dip after going over the bump; this is **overdamped**.

5.27

spring constant  $K = 80 \text{ lb/in}$

mass =  $0.0625 \text{ lb} \cdot \text{s}^2/\text{ft}$

resonant frequency =  $19.7 \text{ Hz}$

## Section 5.4

5.29

a.  $N_G = 4$

b.  $N_G = 2.5$

5.31

$$\theta_2 = 180^\circ$$

$$\theta_2 = 288^\circ$$

5.33

$$N_{\text{Gtot}} = N_{G1} \times N_{G2} \times N_{G3} = 1.75 \times 3 \times 2 = 10.5$$

$$\text{Gear output} = \frac{500 \text{ rpm}}{10.5} = 47.6 \text{ rpm}$$

5.35

With a gear ratio of 7.7, the motor speed would be stepped down to:

$$w_{\text{roller}} = \frac{500 \text{ rpm}}{7.7} = 65 \text{ rpm} \quad \text{motor is OK.}$$

5.37

**Backlash** occurs if there is a space between meshed gear teeth. Then, if one gear is held stationary, the other gear has a small amount of free movement, called “backlash.” Backlash increases in long gear trains, proportional to the gear ratio.

5.39

When the gear ratio of a worm gear is high enough, then **lock-up** occurs. This is when the friction is high enough to prevent “back spinning” of the worm gear by the worm.

5.41

$$N_h = \frac{200}{(200 - 196)} = 50$$

## Section 5.5

5.43

An electromagnetic brake is a device that can slow down or stop a rotating shaft. When activated, a friction disk attached to the rotating shaft is pulled into contact with a stationary disk by an electromagnet, causing a drag on the rotating shaft.

## Section 5.6

5.45

$$\text{fan speed} = 1750 \text{ rpm} \times \frac{3 \text{ in}}{12 \text{ in}} = 438 \text{ rpm}$$

5.47

$$\text{fan speed} = 1750 \text{ rpm} \times \frac{8 \text{ cm}}{30 \text{ cm}} = 467 \text{ rpm}$$

5.49

- Similarities*
1. connect parallel shafts
  2. all power is transmitted on one side
  3. can be used for speed change
- Differences*
1. roller chain is lubricated
  2. roller chain is not tight (does not use friction)
  3. roller chain does not creep

## Section 6.1

6.1

$$V_{\text{pot}} = 4.63 \text{ V}$$

6.3

$$\begin{aligned} \text{Loading error} &= V_{\text{NL}} - V_{\text{L}} = 5 \text{ V} - 4.76 \text{ V} \\ &= 0.24 \text{ V} \end{aligned}$$

6.5

$$\Delta\theta = \frac{\text{error} \times \theta_{\text{tot}}}{100} = \frac{0.25 \times 350^\circ}{100} = 0.875^\circ$$

6.7

- a. maximum shaft rotation =  $\frac{350^\circ}{3} = 117^\circ$
- b. maximum shaft error =  $\frac{0.7^\circ}{3} = 0.23^\circ$

6.9

LSB = 0.0196 V/step (for ADC)

$$\theta_{\text{LSB}} = 1.37^\circ \text{ (without gears)}$$

With 4 : 1 gear ratio:  $\theta_{\text{LSB}} = 0.34^\circ$ ; system is OK.

6.11

$$\frac{360^\circ}{3^\circ/\text{state}} = 120 \text{ states}$$

7 bits  $\rightarrow$  128 states 7 tracks is sufficient

6.13

$$\text{resolution of encoder} = \frac{360^\circ}{500 \text{ slots}} = 0.72^\circ/\text{slot}$$

$$\text{current angle} = \frac{0.72^\circ}{\text{slot}} \times 355 \text{ slots} = 255.6^\circ$$

## Section 6.2

6.15

**Velocity** is the change in position divided by the corresponding change in time:

$$\text{vel} = \frac{d_2 - d_1}{t_2 - t_1}$$

To get velocity from position sensors, take two position readings ( $d_2$  and  $d_1$  at known time interval ( $t_2 - t_1$ )) and apply the formula.

6.17

$$\text{vel} = \frac{d_2 - d_1}{t_2 - t_1} = \frac{135^\circ - 133^\circ}{.25 \text{ s}} = \frac{2^\circ}{0.25 \text{ s}} = 8^\circ/\text{s}$$

6.19

Transfer function for the toothed rotor sensor is:

$$\text{TF} = \frac{18 \text{ Hz}}{1 \text{ rps}} \times \frac{1 \text{ rps}}{60 \text{ rpm}} = \frac{.3 \text{ Hz}}{\text{rpm}}$$

$$\text{Apply TF} \quad 140 \text{ Hz} \times \frac{1 \text{ rpm}}{.3 \text{ Hz}} = 476 \text{ rpm}$$



6.21

$$TF = \frac{2.5 \text{ V}}{1000 \text{ rpm}}$$

$$\text{vel} = \frac{1000 \text{ rpm}}{2.5 \text{ V}} \times 1.5 \text{ V} = 600 \text{ rpm}$$

**Section 6.3**

6.23

1. Switch that detects if car door is closed
2. Switch that detects if refrigerator door is closed

6.25

A slotted coupler contains an LED and photo transistor in one package. Anything (opaque) that moves into the slot will break the light path and be detected. An example would be to detect if a disk is at the correct angle by having the outer edge of the disk pass through the coupler. A hole in the disk would activate the coupler.

**Section 6.4**

6.27

$$k = \frac{\text{Force}}{\text{extension length}} = \frac{180 \text{ lb}}{1.25 \text{ in}} = 144 \text{ lb/in.}$$

6.29

The strain gauge consists of an array of fine wires. If the gauge is extended or compressed, the wires change resistance (slightly). The strain gauge is bonded to a material whose spring constant is known, so that by knowing the elongation of the wires, one can calculate the force that caused the elongation.

6.31

$$F = \rho A = 2400 \frac{\text{lb}}{\text{in}^2} \times 0.785 \text{ in}^2 = 1884 \text{ lb}$$

6.33

$$F = \rho A = 1325 \frac{\text{N}}{\text{cm}^2} \times 3.14 \text{ cm}^2 = 4161 \text{ N}$$

**Section 6.5**

6.35

When a **Bourdon tube** is pressurized it tends to straighten out. The amount that the tube straightens is detected with a position sensor.

6.37

$$\begin{aligned} \text{differential pressure} &= 18.7 \frac{\text{lb}}{\text{in}^2} - 12 \frac{\text{lb}}{\text{in}^2} \\ &= 6.7 \text{ lb/in}^2 \end{aligned}$$

6.39

$$\begin{aligned} \text{differential pressure} &= 121.3 \text{ kPa} - 90 \text{ kPa} \\ &= 31.3 \text{ kPa} \end{aligned}$$

**Section 6.6**

6.41

The **cold junction** of a thermocouple:

1. Is maintained at a constant temperature with a control system.
2. Is compensated for by reading in the ambient temperature and applying a correction factor.
3. Is compensated for by using a temperature-sensitive diode in the thermocouple interface circuit.

6.43

$$V_{\text{net}} = 12 \text{ mV} + 1.5 \text{ mV} = 13.5 \text{ mV}$$

From the graph, 13.5 mV 1 Temp  $\approx$  480°F

6.45

$$\text{Temperature} = 30 \Omega \times \frac{^\circ\text{C}}{0.39 \Omega} = 77^\circ\text{C}$$

**Section 6.7**

6.47

Flow sensors:

- a. orifice plate**—Fluid is forced through a restriction in pipe. Pressure drop across restriction is proportional to flow.

- b. venturi**—Pipe diameter is gently reduced, which speeds up flow and reduces pressure. Flow is proportional to amount of reduced pressure.
- c. Pitot tube**—A small tube faces directly into the flow and detects the impact pressure, which is proportional to flow.
- d. turbin-type**—A small paddle wheel is placed in the flow. Rotational velocity of the wheel is proportional to flow.
- e. magnetic**—For fluids which are at least slightly conductive, a magnetic field is placed around a section of pipe. The moving fluid generates a voltage proportional to flow.

## Section 6.8

6.49

Locate one detector at the 3-ft level and the other at the 4-ft level. When the level drops to 3 ft, turn on the pump. When the level reaches 4 ft, turn off the pump.

6.51

$$P = dH = 9800 \frac{\text{N}}{\text{m}^3} \times 2 \text{ m} = \frac{19,600 \text{ N}}{\text{m}^2} \\ = 19,600 \text{ Pa}$$

6.53

$$H = \frac{P}{d} = 12,000 \frac{\text{N}}{\text{m}^2} \times \frac{\text{m}^3}{6000 \text{ N}} = 2 \text{ m}$$

## Section 6.9

6.55

The software would scan the video “frame” looking for edges. Using the edge information, it would create a new video image in memory of the outline of the part(s). Then, another part of the program would make dimension checks on the outline to determine if the part was OK.

## Section 7.1

7.1

A motor and a generator are fundamentally the same, so when a motor is running, it is also generating an

internal voltage called the **CEMF (counter-EMF)**. This internal voltage is the opposite polarity of the applied voltage, and so it has the effect of reducing the applied voltage. Therefore, as the rpm increases, the CEMF increases, the actual voltage available to the motor decreases, and the torque decreases.

7.3

At 0 rpm:

$$I_A = \frac{V_{\text{in}} - \text{CEMF}}{R_A} = \frac{10 \text{ V} - 0 \text{ V}}{20 \Omega} = 0.5 \text{ A}$$

$$\text{At 1500 rpm: } I_A = \frac{V_{\text{in}} - \text{CEMF}}{R_A} = \frac{10 \text{ V} - 6 \text{ V}}{20 \Omega} \\ = 0.2 \text{ A}$$

## Section 7.2

7.5

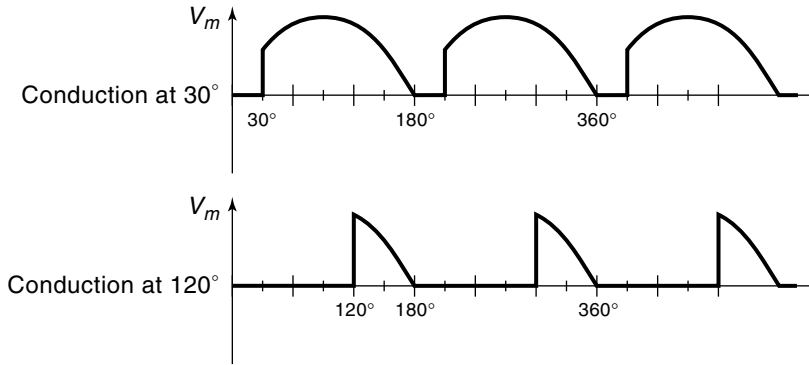
A **series-wound motor** has the armature and field windings connected in series. This motor has a large starting torque and tends to “run away” (go faster and faster) when unloaded.

7.7

A **shunt wound motor** has the armature and field windings connected in parallel. This type of motor has some natural speed regulation, that is, if the load increases, the motor speed tends to stay constant. (Actually, the speed decreases, but not as much as for a series-wound motor.)

7.9

$$\text{Speed regulation} = \frac{S_{\text{NL}} - S_{\text{FL}}}{S_{\text{FL}}} \\ = \frac{2000 - 1750}{1750} \times 100 = 14.3\%$$



### Section 7.3

7.11

The **PM motor** is particularly suited to control applications because it has a straight (linear) torque-speed curve. This simplifies the math operations required of the controller.

7.13

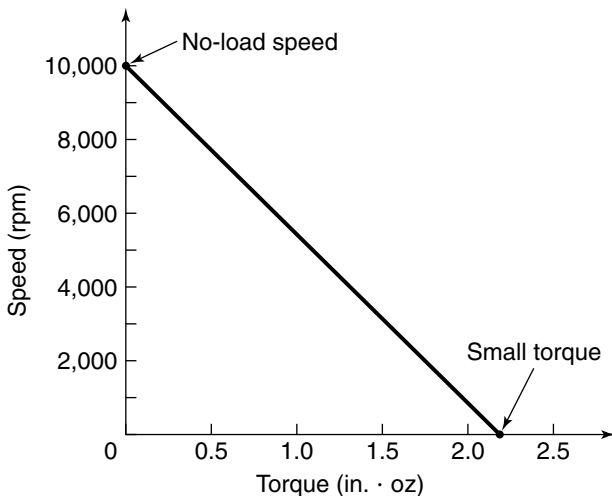
Motor voltage  $\approx 11.5$  V

7.15

No-load speed = 10,100 rpm

Stall torque = 2.19 in. · oz

Voltage = 9 V



### Section 7.4

7.17

A **Class A power amplifier** can be extremely inefficient, because the transistor “absorbs” the power that does not go into the motor. At lower speeds, the transistor may dissipate *more* power than the motor does.

7.19

$$P_{\text{dis}} = VI = 5 \text{ V} \times 1 \text{ A} = 5 \text{ W}$$

7.21

$$\text{Duty cycle} = \frac{1500 \text{ rpm}}{2000 \text{ rpm}} \times 100\% = 75\%$$

7.23

1. PWM is more power efficient, so it doesn't generate as much heat, allowing the components to be smaller and less costly.
2. PWM is controlled by a digital signal, so no DAC may be required.

7.25

(See figure above)

### Section 7.5

7.27

Motor torque = 0.0628 N · m

Motor voltage  $\approx 9$  V

## Section 7.6

### 7.29

1. More reliable
2. More efficient
3. Less maintenance
4. Easily controlled with electronics

## Section 8.1

### 8.1

Final position = 2 complete revolutions + 330° cw

### 8.3

A stepper motor can be **operated open loop** because the controller can reliably know the motor's position by keeping track of the number and direction of steps (providing the motor starts from a known reference position).

### 8.5

Lifting: Weight = 26.7 oz

Holding (power on): Weight = 46.7 oz

Holding (power off): Weight = 6.7 oz

### 8.7

Step	Coil voltages
1	+A -B
2	+A -B and +C -D
3	+C -D
4	-A +B and +C -D
5	-A +B
6	-A +B and -C +D
7	-C +D
8	+A -B and -C +D

## Section 8.2

### 8.9

A **variable reluctance stepper motor** uses a toothed iron wheel for the rotor. When a field pole is energized, the

nearest tooth is pulled into alignment. By having more teeth than poles, the step size can be made quite small.

## Section 8.3

### 8.11

The **hybrid stepper motor** combines the features of the PM and VR stepper motors. It has a permanent magnet between two toothed iron rotors. The magnetic field from the permanent magnet gives the motor a detent torque.

## Section 8.4

### 8.13

Position	Pole 1	Pole 2
1	5 V	0 V
2	4.5 V	0.5 V
3	4 V	1 V
4	3.5 V	1.5 V
5	3 V	2 V
6	2.5 V	2.5 V
7	2 V	3 V
8	1.5 V	3.5 V
9	1 V	4 V
10	0.5 V	4.5 V

### 8.15

**Bilevel drive** is a technique for improving the torque at higher stepping rates. At the beginning of each step, a higher voltage is applied to the motor coil, which causes a steep current rise. After the current is established, a lower voltage is switched in for the remainder of the step period.

## Section 8.5

### 8.17

- a. Steps to advance 1.25 in. = 6000 steps
- b. Linear advance per step = 0.000208 in./step

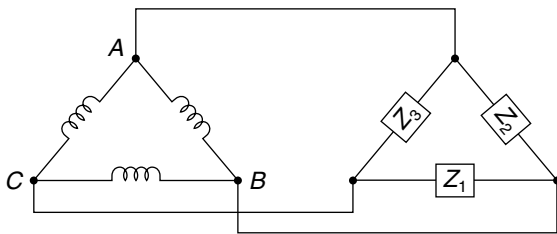
**Section 9.1**

9.1

- a. -170 V
- b. 0 V
- c. 0 V

9.3

A delta-connection generator and load

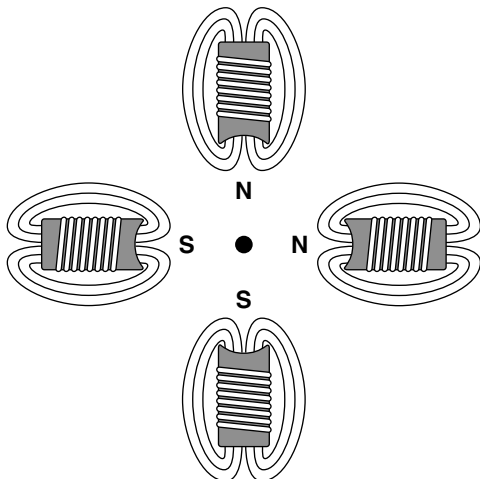


9.5

$$E = \sqrt{3} V_p = 1.732 \times 240 \text{ V} = 416 \text{ Vac}$$

**Section 9.2**

9.7



9.9

$$P = \frac{120f}{S_s} = \frac{120 \times 60}{1200} = 6 \text{ pole motor}$$

9.11

$$S_s = \frac{120f}{P} = \frac{120 \times 400}{8} = 6000 \text{ rpm}$$

8 poles per phase, and 1 phase → 8 individual poles

9.13

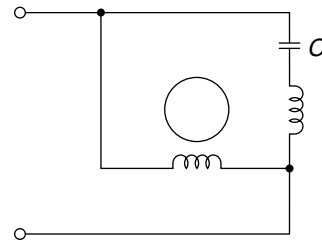
$$S_{\text{Rotor}} = S_s - S_s (\text{Slip}) = 1200 - 0.05(1200) = 1140 \text{ rpm}$$

9.15

The *three phase motor is reversible*, by reversing any two of the three phase voltages.

9.17

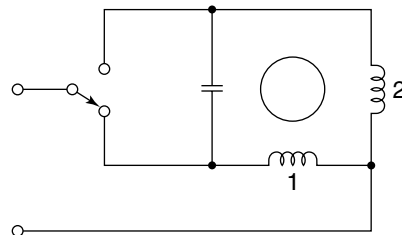
Split-phase motor:



The split-phase motor has two windings: one connected directly across the power lines, and one connected through a capacitor, which phase shifts the power. These two windings provide a true rotating field, and so the motor is self-starting. In essence, it is a two-phase motor running on single-phase power.

9.19

Split-phase control motor:



This motor is bidirectional. If the switch is down, winding 1 gets the power directly, and winding 2 is phase-shifted. When the switch is up, winding 2 gets the power directly and winding 1 is phase-shifted.

### Section 9.3

#### 9.21

1. The synchronous motor runs at synchronous speed.
2. The synchronous motor has slip rings to get power to the rotor.
3. The synchronous motor is not self-starting (even 3-phase type).

#### 9.23

1. Cause the motor frequency to start slow and then speed up; the motor will follow the rotating field.
2. Use another motor, called a pony motor, to get the synchronous motor up to speed.
3. Insert some “squirrel cage” rotor bars in the rotor. The motor starts as an induction motor.

### Section 9.5

#### 9.25

Reversing circuit for 3-phase motor  
(See figure at top of next page)

#### 9.27

1. Changing the voltage will change the speed somewhat, because a reduced voltage will increase the slip.
2. Changing the frequency will change the speed, because it changes the speed of the rotating field. The frequency can be changed with a circuit such as a DC link converter.

### Section 10.1

#### 10.1

A leadscrew linear actuator works as follows: an electric motor turns a threaded shaft. The output motion is

taken from a “nut,” which takes a linear motion along the threaded shaft.

Leadscrew linear activators are less messy and don’t require additional hardware (pumps, tubing) that hydraulic systems do.

Leadscrew linear-actuators move slower than hydraulic systems, and can’t store up energy for short, big, bursts of power.

#### 10.3

$$\text{Time} = 12 \text{ in.} \times \frac{1 \text{ s}}{0.85 \text{ in.}} = 14.1 \text{ s}$$

#### 10.5

Type 885 meets the specifications.

### Section 10.2

#### 10.7

$$F = P \times A = 500 \frac{\text{lb}}{\text{in}^2} \times 12.57 \text{ in}^2 = 6283 \text{ lb}$$

#### 10.9

$$F = 27,475 \text{ N}$$

#### 10.11

$$\text{Diameter} = 3.0 \text{ in.}$$

#### 10.13

$$F = PA = \frac{104 \text{ lb}}{\text{in}^2} = 12.56 \text{ in}^2 = 1306 \text{ lb}$$

#### 10.15

See Figure 10.22.

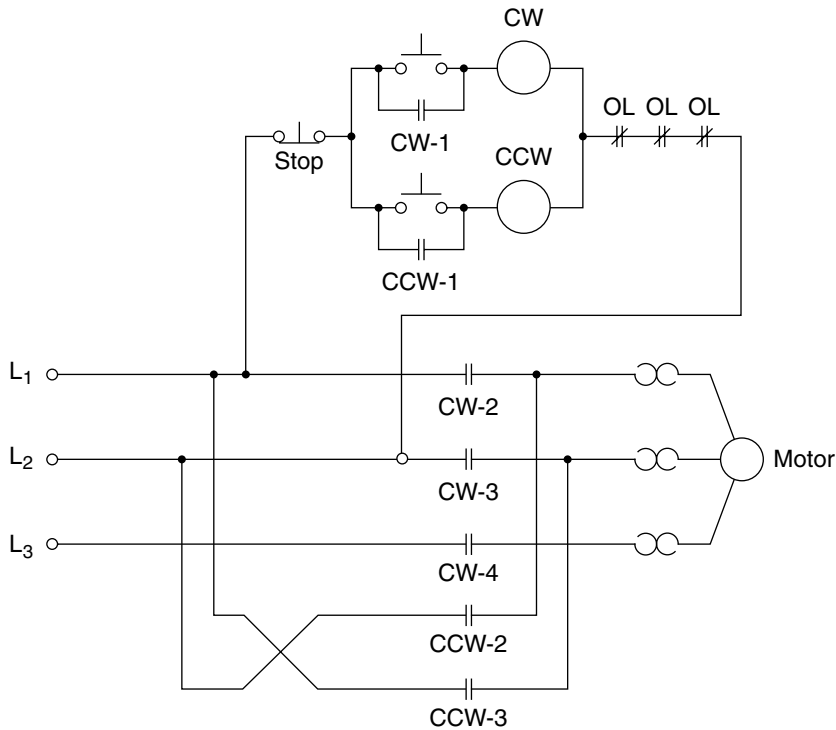
### Section 10.3

#### 10.17

$$F = PA = \frac{100 \text{ lb}}{\text{in}^2} \times 0.785 \text{ in}^2 = 78.5 \text{ lb}$$

#### 10.19

$$\text{Piston diameter} = 0.92 \text{ in.}$$



10.21

Piston force left over (after spring):  $35.4 \text{ lb} - 12 \text{ lb} = 23.4 \text{ lb}$

## Section 11.1

11.1

Rise time: 1.0  
Overshoot: 0.7  
Settling time: 6

## Section 11.2

11.3

The level of liquid in a tank is to be maintained at 30 in. deep. A level detector at 29 in. turns on the pump. Another level detector at 31 in. turns off the pump.

11.5

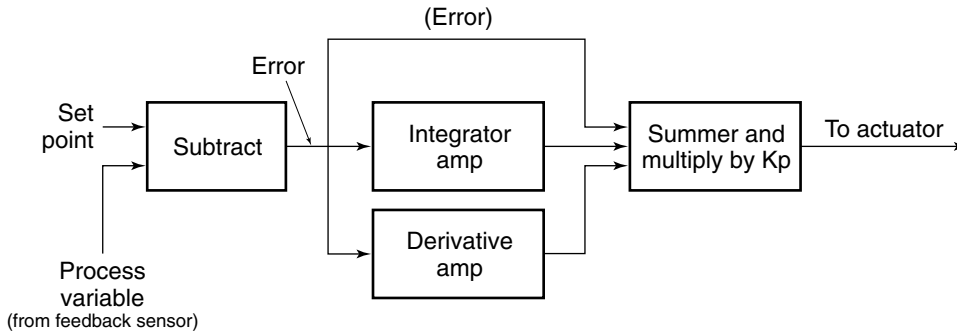
A wind direction sensor mounted on the windmill indicates if the wind direction is more than say,  $20^\circ$  either side of straight on. If the wind is coming from the right (more than  $20^\circ$ ), the motor comes on and rotates the windmill CW. If the wind is coming from the left (more than  $20^\circ$ ), the motor comes on and rotates the windmill CCW (until it's within  $20^\circ$ ).

## Section 11.3

11.7

Torque when arm is  $15^\circ = 70 \text{ ft} \cdot \text{lb}$

Torque when arm is  $45^\circ = 10 \text{ ft} \cdot \text{lb}$



### 11.9

A simple proportional system can never reduce the position error to zero when lifting a weight. This is because the proportional system *only* provides a force when there *is* an error. Therefore, to provide the lifting force, there would have to be an error (the weight would sag an amount proportional to its weight).

### 11.11

Error =  $\pm 0.05$  in.

Total dead band =  $2E = 0.1$  in.

## Section 11.4

### 11.13

Integral feedback will provide increasing restoring force *as long as there is any error*. Therefore, if a system has steady-state error, the restoring force will continue to increase until the error is eliminated.

### 11.15

$$\text{Output} = K_p K_I \sum (E \Delta t)$$

For steady state error, Output =  $K_p K_I E \cdot t$

$$t = \frac{\text{Output}}{K_p K_I E} = \frac{20 \text{ in} \cdot \text{lb}}{(4 \text{ in} \cdot \text{lb}/^\circ)(2/s)(5^\circ)} = .5\text{s}$$

## Section 11.5

### 11.17

Derivative feedback produces a restoring force which is proportional to the rate-of-change of the error, that

is, if the error is changing rapidly, the derivative control provides a large force, and if the error is changing slowly, a small force.

### 11.19

Derivative feedback makes a control system more responsive to rapid change, because a quick change will have a large error rate-of-change, which produces a large, immediate restoring force. Derivative feedback reduces overshoot because it “applies the brakes” *before* reaching the set point. This occurs because, as the controlled object slows, it has a negative error rate-of-change, which results in a negative restoring force (brakes).

## Section 11.6

### 11.21

Block diagram of an analog PID controller. (See figure above)

### 11.23

The digital controller *integrates* the error signal by evaluating a “difference equation.”

$$K_I \sum (E \Delta t) = K_I E_1 T + K_I E_2 T + K_I E_3 T + \dots$$

In other words, the controller accumulates the area under the error curve as rectangles, one new rectangle for each new sample.

### 11.25

BASIC program to implement a PID controller.



$$K_P = 2 \quad K_I = 1.5 \quad K_D = 1.8$$

Assume the following port adrs SP = 100, PV = 120,  
OUTPID = 140

```

OUTI = 0      Initialize to 0
OLDDV = 0    Initialize to 0
INP 100, SP   Read in set point
INP 120, PV   Read in feedback sensor
E = SP - PV   Calculate error
OUTI = OUTI + (1.5 * E * .5)  Calculate output
                                from integral
                                control
NEWDV = 1.8 * E
OUTD = (NEWDV - OLDDV) / .5 } Calculate output
                                from derivative
                                control
OLDDV = NEWDV
OUTPID = 2 * (E + OUTI + OUTD) Calculate total
                                PID output
OUT 140, OUTPID Send output to
                                actuator

```

Program would wait  $\approx .5$  s with a time delay loop, then loop back

### 11.27

$$\text{Lag} = 72^\circ$$

### 11.29

$$N = \frac{\Delta PV}{T} = \frac{6\%}{1.25 \text{ s}} = \frac{4.8\%}{\text{s}}$$

$$K_P = \frac{1.2 \Delta CV}{NL} = \frac{1.2 \times 8\% \text{ s}}{4.8\% \times 0.7 \text{ s}} = 2.9$$

$$K_I = \frac{1}{2L} = \frac{1}{2 \times 0.7 \text{ s}} = .7/\text{s}$$

$$K_D = 0.5 L = 0.5 \times 0.7 \text{ s} = 0.35 \text{ s}$$

### 11.31

A high sample rate is good because it allows the controller to know more accurately what the controlled vari-

able is doing, in particular, the controller can respond more quickly to any rapid changes in the system.

### 11.33

“Feedforward” is when the controller knows in advance the path it should take. This allows the controller to make better decisions about when to slow down and when to speed up. A PIP controller uses feedforward.

## Section 11.7

### 11.35

Rule 2 (gas OK) applies 26%.

Rule 1 (turn gas up) applies 33%.

Turn gas up 2.53 notches.

### 11.37

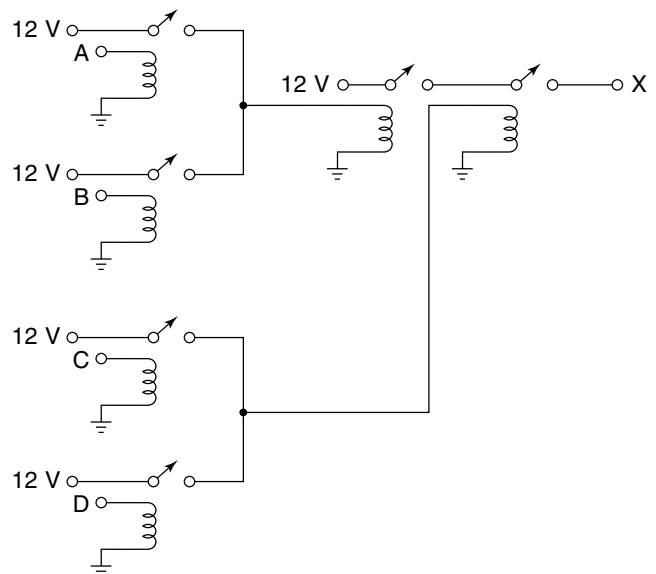
For temperature =  $62^\circ$ : Cool = 33%, Medium = 26%

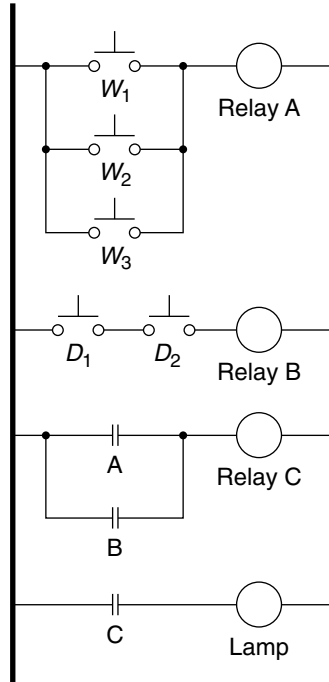
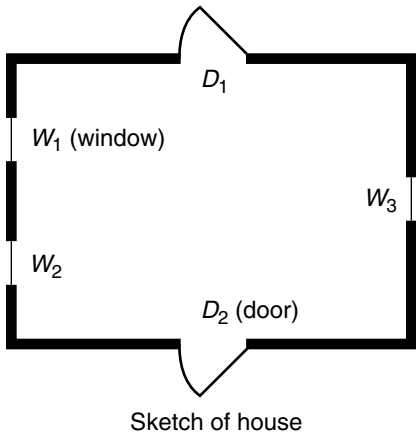
For  $\Delta T = +.7^\circ/\text{min}$ : Steady = 0%, Rising = 60%

Weighted mean output =  $-0.137$  notches

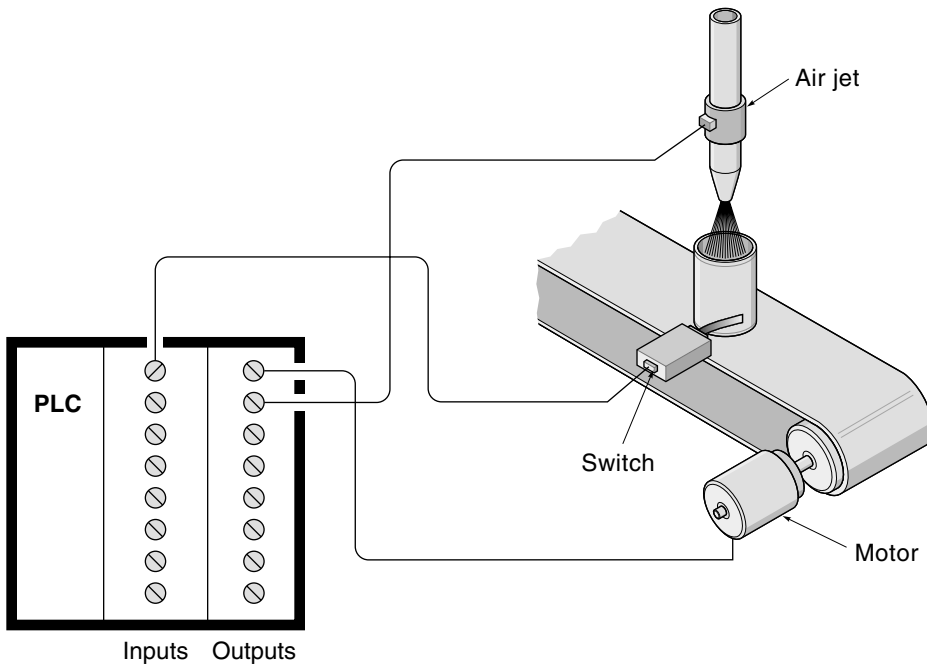
## Section 12.1

### 12.1





**Exercise 12.3**



**Exercise 12.7**

## 12.3

(See figure at top of previous page)

## Section 12.2

## 12.5

Installing a PLC

*Hardware*

1. Decide where the PLC will be located and install cables to I/O devices.
2. Connect input devices (switches, sensors, etc.) to input port connectors on PLC.
3. Connect output devices (motors, relays, etc.) to output port connectors on PLC.

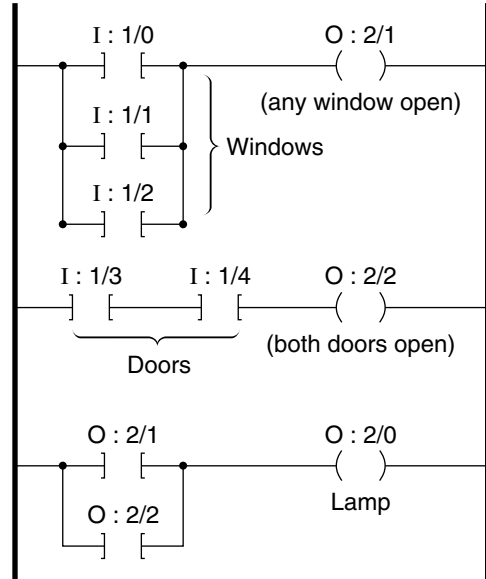
*Software*

1. Develop a ladder diagram of the program.
2. Using software (from the PLC manufacturer), enter the program into a PC.
3. Connect the PC to the PLC and download the program into the PLC.
4. Test the operation of the program, make modifications as required.
5. When done, disconnect the PC and allow the PLC to operate on its own.

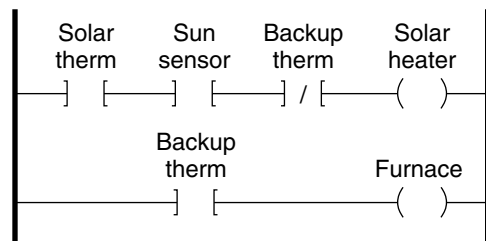
## 12.7

(See figure at bottom of previous page)

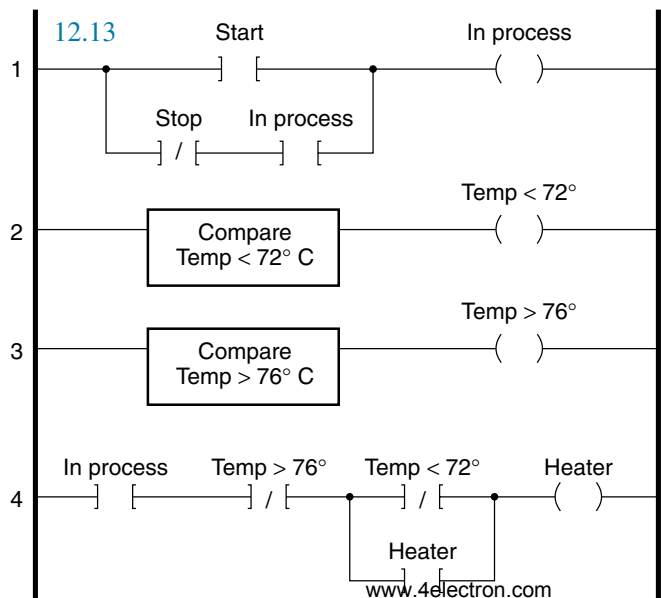
## 12.9



## 12.11



## 12.13



### 12.15

A device level network connects sensors and motors to a PLC (or controller) and may bridge to other networks throughout the facility. The advantages are:

1. Less wiring. Only one cable is required, all the devices tap into the cable.
2. Data is transmitted digitally with error checking, so there is no signal attenuation.
3. Devices tend to be more “intelligent” because if they are on the net, they have a way to report such things as their condition *in addition* to their primary function.

### 12.17

#1->1000X	Assign motor #1 to X-axis
#2->1000Y	Assign motor #2 to Y-axis
#3->1000Z	Assign motor #3 to Z-axis
CLEAR	
LINEAR	Specify motion mode (trapezoid)
TA500	Set acceleration
TS200	Set speed
X0 Y0 Z0	Move to start position
X5 Y5 Z2	Move to first position
X4 Y4 Z3	Move to next position
DWELL	Wait 500ms
X0 Y0 Z0	Move to start position