



Chapter 3 Data and Signals

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To be transmitted, data must be transformed to electromagnetic signals.

Data can be analog or digital. The term analog data refers to information that is <u>continuous</u>; digital data refers to information that has <u>discrete</u> states. Analog data take on continuous values. Digital data take on discrete values.

Topics discussed in this section:

Analog and Digital Data Analog and Digital Signals Periodic and Nonperiodic Signals

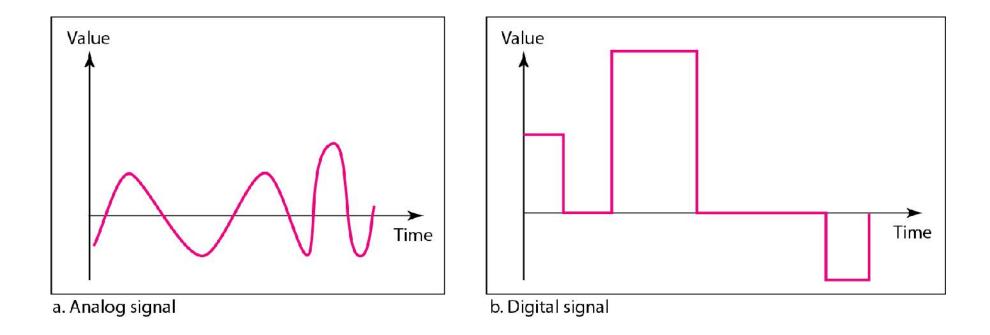


Data can be analog or digital. Analog data are continuous and take continuous values. Digital data have discrete states and take discrete values.



Signals can be analog or digital. Analog signals can have an <u>infinite</u> number of values in a range; digital signals can have only a <u>limited</u> number of values.

Figure 3.1 Comparison of analog and digital signals





In data communications, we commonly use periodic analog signals and nonperiodic digital signals.

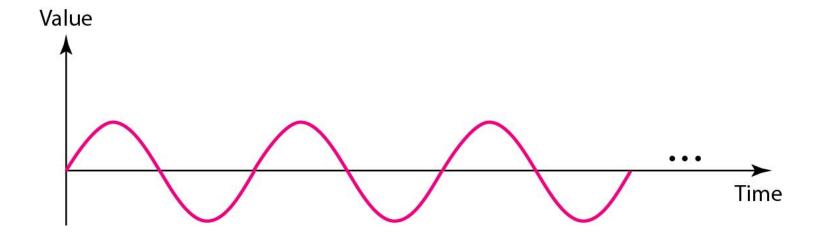
- A periodic signal completes a pattern within a measurable time frame, called a period, and <u>repeats</u> that pattern over subsequent identical periods.
- The completion of one full pattern is called a <u>cycle</u>.
- A nonperiodic signal changes without exhibiting a pattern or cycle that repeats over time.

3-2 PERIODIC ANALOG SIGNALS

Periodic analog signals can be classified as simple or composite. A simple periodic analog signal, a sine wave, cannot be decomposed into simpler signals. A composite periodic analog signal is composed of multiple sine waves.

Topics discussed in this section:

Sine Wave Wavelength Time and Frequency Domain Composite Signals Bandwidth Figure 3.2 A sine wave







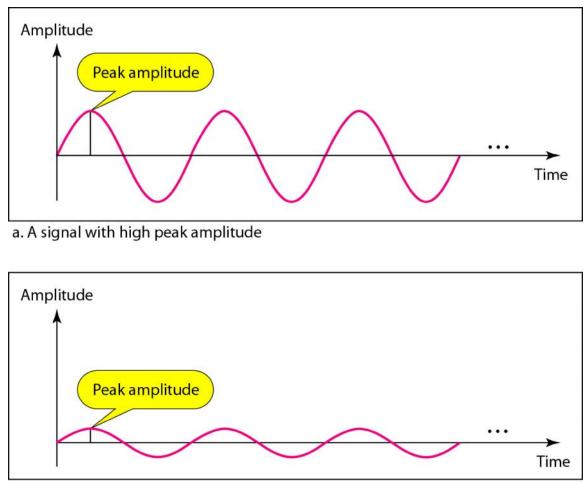
We discuss a mathematical approach to sine waves in Appendix C.

Example 3.1

The power in your house can be represented by a sine wave with a peak amplitude of 155 to 170 V. However, it is common knowledge that the voltage of the power in U.S. homes is 110 to 120 V. This discrepancy is due to the fact that these are **root mean square** (rms) values. The signal is squared and then the average amplitude is calculated. The peak value is equal to $2^{\frac{1}{2}} \times rms$ value.

$$V_p = \sqrt{2} x V_{rms}$$

Figure 3.3 Two signals with the same phase and frequency, but different amplitudes



b. A signal with low peak amplitude

Example 3.2

The voltage of a battery is a constant; this constant value can be considered a sine wave, as we will see later. For example, the peak value of an AA battery is normally 1.5 V.

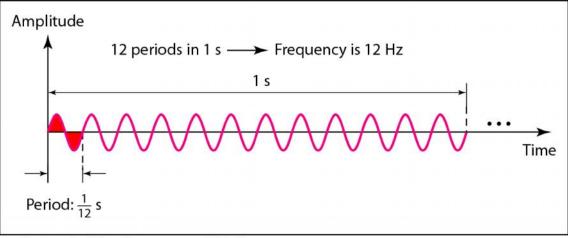




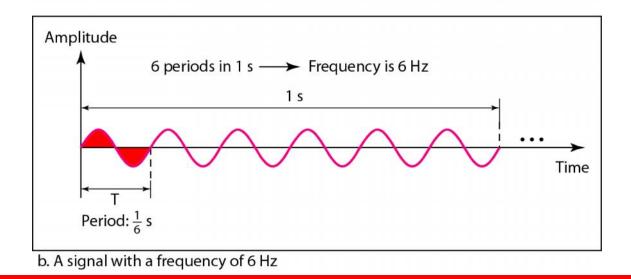
Frequency and period are the inverse of each other.

$$f = \frac{1}{T}$$
 and $T = \frac{1}{f}$

Figure 3.4 Two signals with the same amplitude and phase, but different frequencies



a. A signal with a frequency of 12 Hz



Unit	Equivalent	Unit	Equivalent
Seconds (s)	1 s	Hertz (Hz)	1 Hz
Milliseconds (ms)	10^{-3} s	Kilohertz (kHz)	10^3 Hz
Microseconds (µs)	10 ⁻⁶ s	Megahertz (MHz)	10 ⁶ Hz
Nanoseconds (ns)	10 ⁻⁹ s	Gigahertz (GHz)	10 ⁹ Hz
Picoseconds (ps)	10 ⁻¹² s	Terahertz (THz)	10 ¹² Hz

Table 3.1Units of period and frequency



The power we use at home has a frequency of 60 Hz. The period of this sine wave can be determined as follows:

$$T = \frac{1}{f} = \frac{1}{60} = 0.0166 \text{ s} = 0.0166 \times 10^3 \text{ ms} = 16.6 \text{ ms}$$

Example 3.4

Express a period of 100 ms in microseconds.

Solution

From Table 3.1 we find the equivalents of 1 ms (1 ms is 10^{-3} s) and 1 s (1 s is 10^{6} µs). We make the following substitutions:.

 $100 \text{ ms} = 100 \times 10^{-3} \text{ s} = 100 \times 10^{-3} \times 10^{6} \text{ } \mu\text{s} = 10^{2} \times 10^{-3} \times 10^{6} \text{ } \mu\text{s} = 10^{5} \text{ } \mu\text{s}$

Example 3.5

The period of a signal is 100 ms. What is its frequency in kilohertz?

Solution

First we change 100 ms to seconds, and then we calculate the frequency from the period (1 Hz = 10^{-3} kHz).

$$100 \text{ ms} = 100 \times 10^{-3} \text{ s} = 10^{-1} \text{ s}$$

 $f = \frac{1}{T} = \frac{1}{10^{-1}} \text{ Hz} = 10 \text{ Hz} = 10 \times 10^{-3} \text{ kHz} = 10^{-2} \text{ kHz}$



Frequency is the rate of change with respect to time.

Change in a <u>short span</u> of time means <u>high frequency</u>.

Change over a long span of time means low frequency.





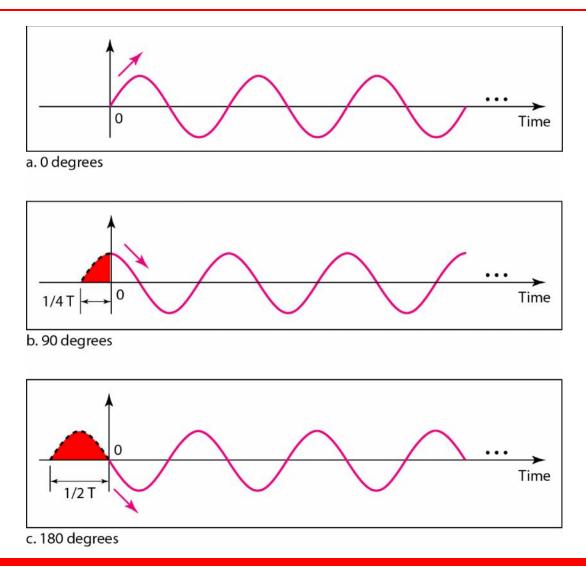
If a signal does not change at all, its frequency is zero. If a signal changes instantaneously, its frequency is infinite.





Phase describes the position of the waveform relative to time 0.

Figure 3.5 Three sine waves with the same amplitude and frequency, but different phases





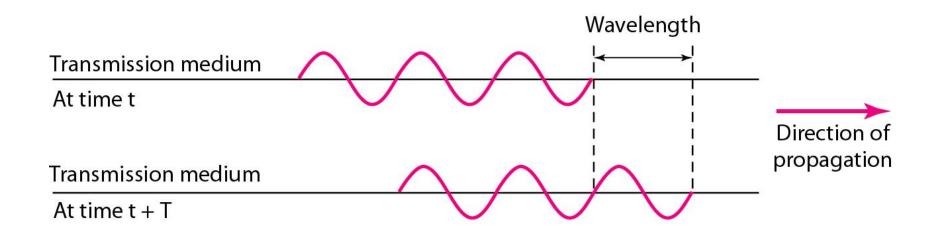
A sine wave is offset 1/6 cycle with respect to time 0. What is its phase in degrees and radians?

Solution

We know that 1 complete cycle is 360[•]. Therefore, 1/6 cycle is

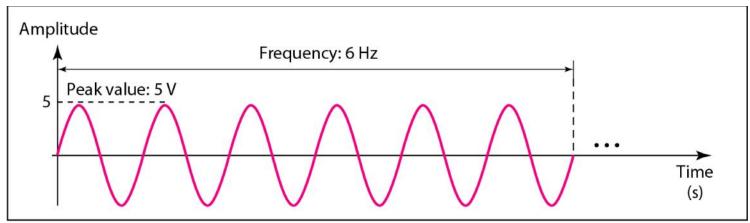
$$\frac{1}{6} \times 360 = 60^\circ = 60 \times \frac{2\pi}{360}$$
 rad $= \frac{\pi}{3}$ rad $= 1.046$ rad

Figure 3.6 Wavelength and period

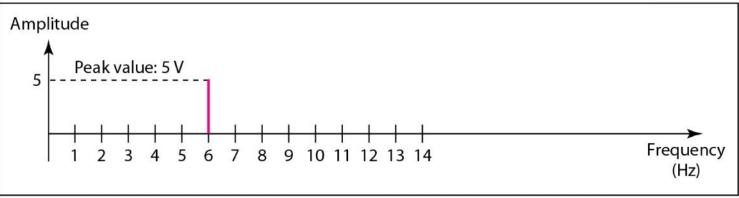


- While the frequency of a signal is independent of the medium, the wavelength depends on both the frequency and the medium.
- Wavelength is a property of any type of signal.
- In data communications, we often use wavelength to describe the transmission of light in an optical fiber.
- The **wavelength** is the <u>distance a simple signal can travel in one period</u>.

Figure 3.7 The time-domain and frequency-domain plots of a sine wave



a. A sine wave in the time domain (peak value: 5 V, frequency: 6 Hz)



b. The same sine wave in the frequency domain (peak value: 5 V, frequency: 6 Hz)



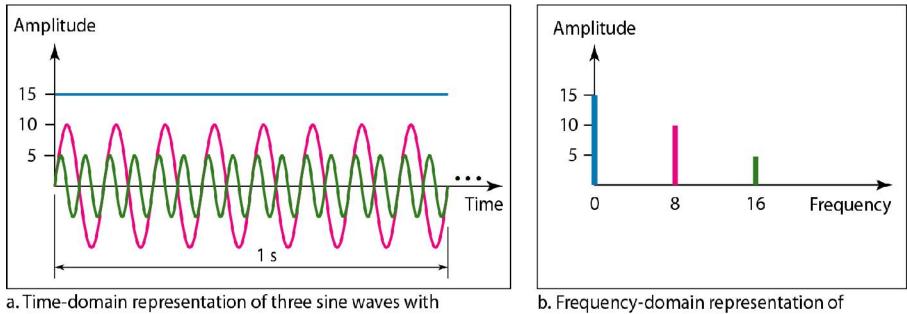


A complete sine wave in the time domain can be represented by one single spike in the frequency domain.

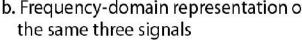
Example 3.7

The <u>frequency domain</u> is more compact and useful when we are dealing with more than one sine wave. For example, Figure 3.8 shows three sine waves, each with different amplitude and frequency. All can be represented by three spikes in the frequency domain.

Figure 3.8 The time domain and frequency domain of three sine waves



frequencies 0, 8, and 16





A single-frequency sine wave is not useful in data communications; we need to send a composite signal, a signal made of many simple sine waves.



According to Fourier analysis, any composite signal is a combination of simple sine waves with different frequencies, amplitudes, and phases. Fourier analysis is discussed in Appendix C.



If the composite signal is periodic, the decomposition gives a series of signals with <u>discrete frequencies;</u> if the composite signal is nonperiodic, the decomposition gives a combination of sine waves with <u>continuous</u> <u>frequencies</u>.

Example 3.8

Figure 3.9 shows a periodic composite signal with frequency f. This type of signal is not typical of those found in data communications. We can consider it to be three alarm systems, each with a different frequency. The analysis of this signal can give us a good understanding of how to decompose signals. **Figure 3.9** A composite periodic signal

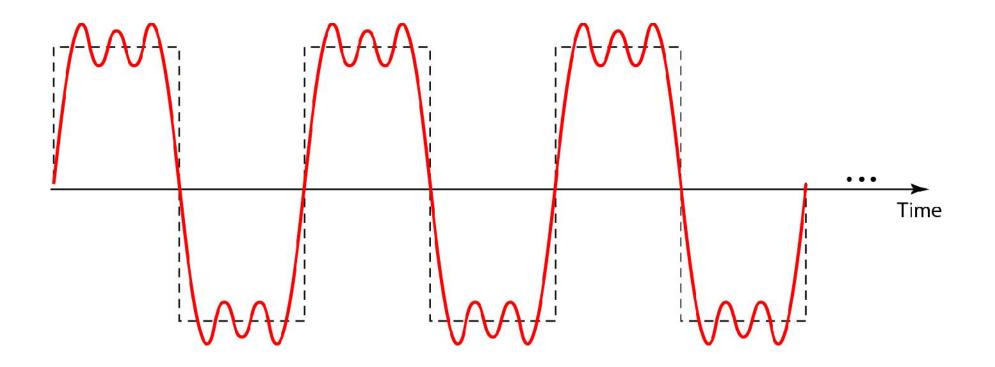
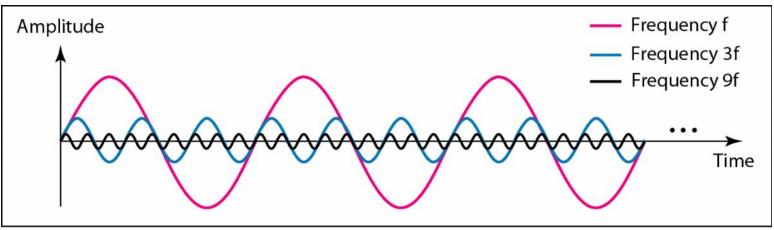
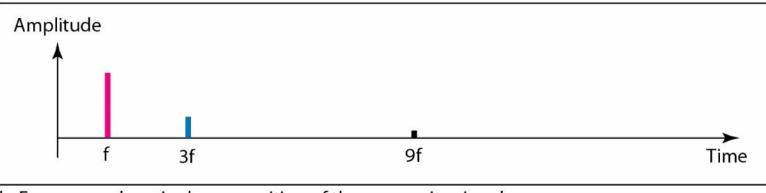


Figure 3.10 Decomposition of a composite periodic signal in the time and frequency domains



a. Time-domain decomposition of a composite signal

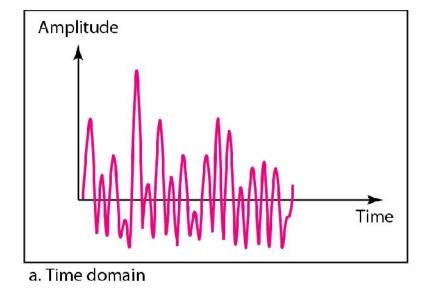


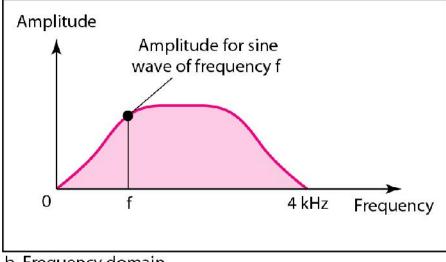
b. Frequency-domain decomposition of the composite signal

Example 3.9

Figure 3.11 shows a nonperiodic composite signal. It can be the signal created by a microphone or a telephone set when a word or two is pronounced. In this case, the composite signal cannot be periodic, because that implies that we are repeating the same word or words with exactly the same tone.

Figure 3.11 The time and frequency domains of a nonperiodic signal





b. Frequency domain

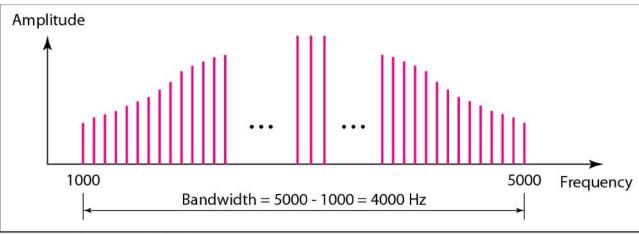


The bandwidth of a composite signal is the difference between the highest and the lowest frequencies contained in that signal.

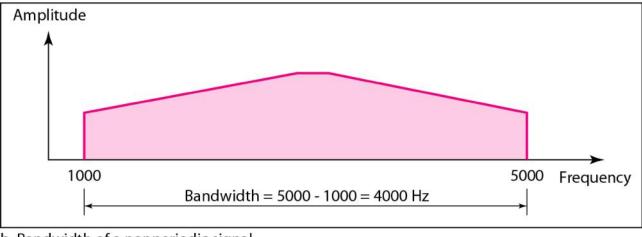
$$\boldsymbol{B} = \boldsymbol{f}_H - \boldsymbol{f}_L$$

3.38

Figure 3.12 The bandwidth of periodic and nonperiodic composite signals



a. Bandwidth of a periodic signal



b. Bandwidth of a nonperiodic signal

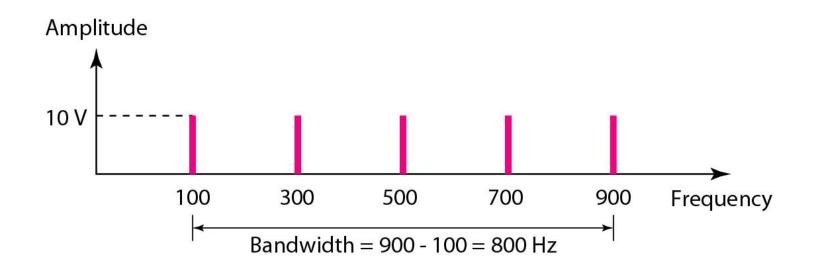
If a periodic signal is decomposed into five sine waves with frequencies of 100, 300, 500, 700, and 900 Hz, what is its bandwidth? Draw the spectrum, assuming all components have a maximum amplitude of 10 V. <u>Solution</u>

Let f_h be the highest frequency, f_l the lowest frequency, and **B** the bandwidth. Then

$$B = f_h - f_l = 900 - 100 = 800 \text{ Hz}$$

The spectrum has only five spikes, at 100, 300, 500, 700, and 900 Hz (see Figure 3.13).

Figure 3.13 The bandwidth for Example 3.10



A periodic signal has a bandwidth of 20 Hz. The highest frequency is 60 Hz. What is the lowest frequency? Draw the spectrum if the signal contains all frequencies of the same amplitude.

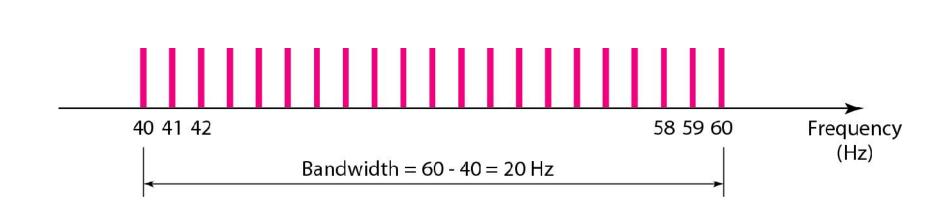
Solution

Let f_h be the highest frequency, f_l the lowest frequency, and **B** the bandwidth. Then

 $B = f_h - f_l \implies 20 = 60 - f_l \implies f_l = 60 - 20 = 40 \text{ Hz}$

The spectrum contains all integer frequencies. We show this by a series of spikes (see Figure 3.14).

Figure 3.14 The bandwidth for Example 3.11

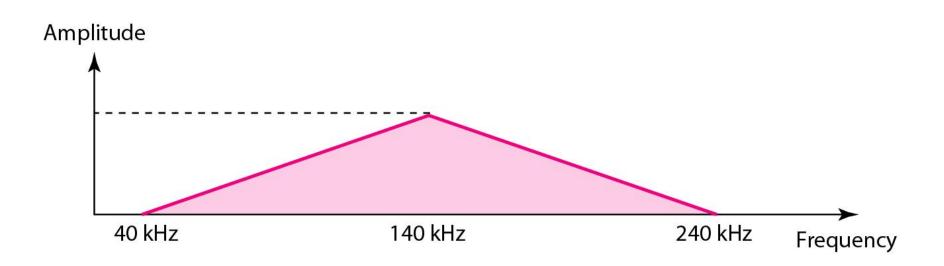


A nonperiodic composite signal has a bandwidth of 200 kHz, with a middle frequency of 140 kHz and peak amplitude of 20 V. The two extreme frequencies have an amplitude of 0. Draw the frequency domain of the signal.

Solution

The lowest frequency must be at 40 kHz and the highest at 240 kHz. Figure 3.15 shows the frequency domain and the bandwidth.

Figure 3.15 The bandwidth for Example 3.12



An example of a nonperiodic composite signal is the signal propagated by an AM radio station. In the United States, each AM radio station is assigned a 10-kHz bandwidth. The total bandwidth dedicated to AM radio ranges from 530 to 1700 kHz. We will show the rationale behind this 10-kHz bandwidth in Chapter 5.

Another example of a nonperiodic composite signal is the signal propagated by an FM radio station. In the United States, each FM radio station is assigned a 200kHz bandwidth. The total bandwidth dedicated to FM radio ranges from 88 to 108 MHz. We will show the rationale behind this 200-kHz bandwidth in Chapter 5.

Example 3.15

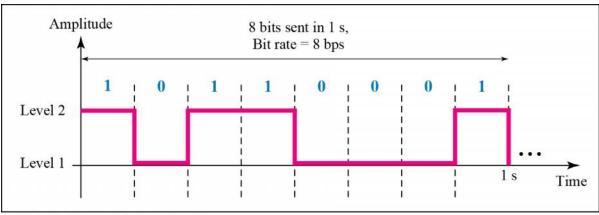
Another example of a nonperiodic composite signal is the signal received by an old-fashioned analog blackand-white TV. A TV screen is made up of pixels. If we assume a resolution of 525×700 , we have 367,500pixels per screen. If we scan the screen 30 times per second, this is $367,500 \times 30 = 11,025,000$ pixels per second. The worst-case scenario is alternating black and white pixels. We can send 2 pixels per cycle. Therefore, we need 11,025,000 / 2 = 5,512,500 cycles per second, or Hz. The bandwidth needed is 5.5125 MHz.

In addition to being represented by an analog signal, information can also be represented by a digital signal. For example, a 1 can be encoded as a positive voltage and a 0 as zero voltage. A digital signal can have more than two levels. In this case, we can send more than 1 bit for each level.

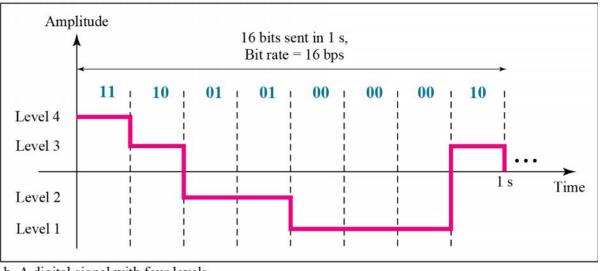
Topics discussed in this section:

Bit Rate Bit Length Digital Signal as a Composite Analog Signal Application Layer

Figure 3.16 Two digital signals: one with two signal levels and the other with four signal levels



a. A digital signal with two levels



b. A digital signal with four levels





Appendix C reviews information about exponential and logarithmic functions.

A digital signal has eight levels. How many bits are needed per level? We calculate the number of bits from the formula

Number of bits per level = $\log_2 8 = 3$

Each signal level is represented by 3 bits.

No. of level, $L = 2^n$ where n = no. of bit

A digital signal has nine levels. How many bits are needed per level?

We calculate the number of bits by using the formula. $n = \log_2 9$

Each signal level is represented by 3.17 bits. However, this answer is not realistic. The number of bits sent per level needs to be an integer as well as a power of 2. For this example, 4 bits can represent one level.

$$L = 2^{n}$$

or, $n = \log_2 L$

Assume we need to download text documents at the rate of 100 pages per minute. What is the required bit rate of the channel?

Solution

A page is an average of 24 lines with 80 characters in each line. If we assume that one character requires 8 bits, the bit rate is

 $100 \times 24 \times 80 \times 8 = 1,636,000$ bps = 1.636 Mbps

The bit rate is the number of bits sent in 1s, expressed in bits per second (bps)

A digitized voice channel, as we will see in Chapter 4, is made by digitizing a 4-kHz bandwidth analog voice signal. We need to sample the signal at twice the highest frequency (two samples per hertz). We assume that each sample requires 8 bits. What is the required bit rate?

Solution

The bit rate can be calculated as

 $2 \times 4000 \times 8 = 64,000$ bps = 64 kbps

$$fs = 2 x f$$

What is the bit rate for high-definition TV (HDTV)?

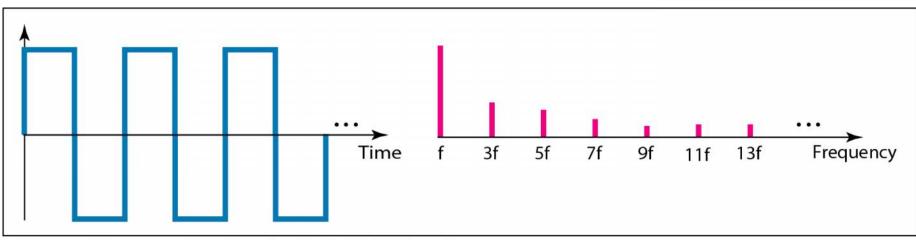
Solution

HDTV uses digital signals to broadcast high quality video signals. The HDTV screen is normally a ratio of 16:9. There are 1920 by 1080 pixels per screen, and the screen is renewed 30 times per second. Twenty-four bits represents one color pixel.

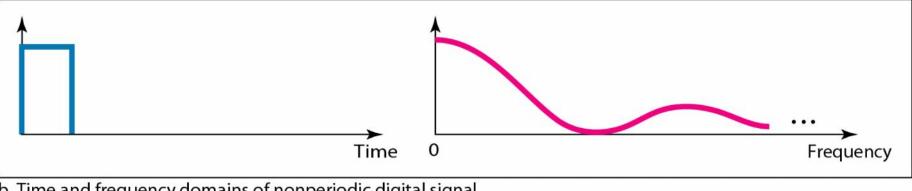
 $1920 \times 1080 \times 30 \times 24 = 1,492,992,000$ or 1.5 Gbps

The TV stations reduce this rate to 20 to 40 Mbps through compression.

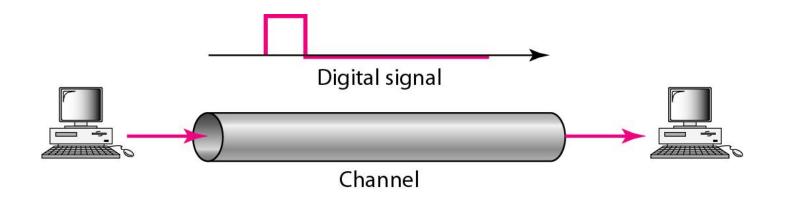
Figure 3.17 The time and frequency domains of <u>periodic</u> and <u>nonperiodic</u> digital signals



a. Time and frequency domains of periodic digital signal



b. Time and frequency domains of nonperiodic digital signal



How can we send a digital signal from point A to point B? We can transmit a digital signal by using one of two different approaches:

baseband transmission or

broadband transmission (using modulation).

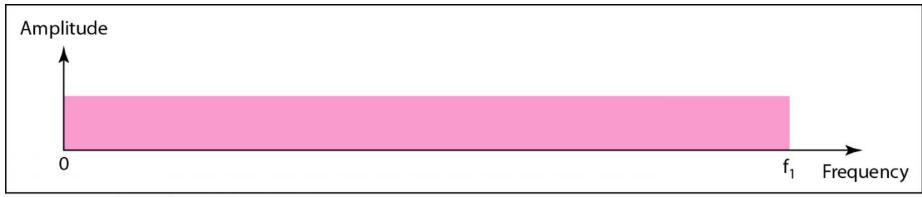
Baseband transmission means sending a digital signal over a channel <u>without</u> <u>changing</u> the digital signal to an analog signal.



A digital signal is a composite analog signal with an infinite bandwidth.

- Baseband transmission requires that we have a <u>low-pass</u> <u>channel</u>, a channel with a bandwidth that starts from zero.
- This is the case if we have a dedicated medium with a bandwidth constituting only one channel.

Figure 3.19 Bandwidths of two low-pass channels

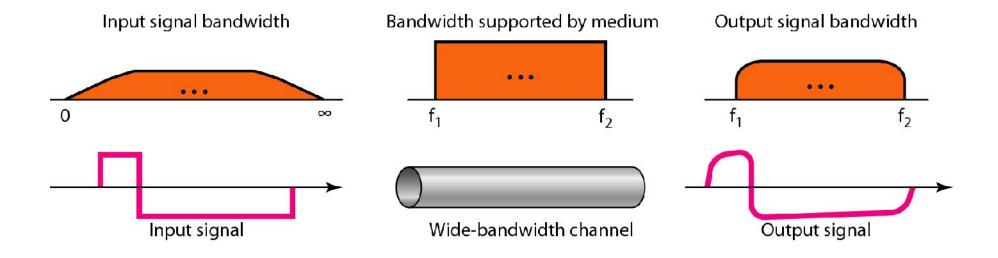


a. Low-pass channel, wide bandwidth



b. Low-pass channel, narrow bandwidth

Figure 3.20 Baseband transmission using a dedicated medium



- Although the output signal is not an exact replica of the original signal, the data can still be deduced from the received signal.
- Although some of the frequencies are blocked by the medium, they are not critical.



Baseband transmission of a digital signal that preserves the shape of the digital signal is possible only if we have a low-pass channel with an infinite or very wide bandwidth.

Example 3.21

An example of a dedicated channel where the entire bandwidth of the medium is used as one single channel is a LAN. Almost every wired LAN today uses a dedicated channel for two stations communicating with each other. In a bus topology LAN with multipoint connections, only two stations can communicate with each other at each moment in time (timesharing); the other stations need to refrain from sending data. In a star topology LAN, the entire channel between each station and the hub is used for communication between these two entities. We study LANs in Chapter 14.

Low-Pass Channel with Limited Bandwidth

- In a low-pass channel with limited bandwidth, we <u>approximate</u> the digital signal with an analog signal.
- The level of approximation depends on the bandwidth available.

Figure 3.21 Rough approximation of a digital signal using the first harmonic for worst case

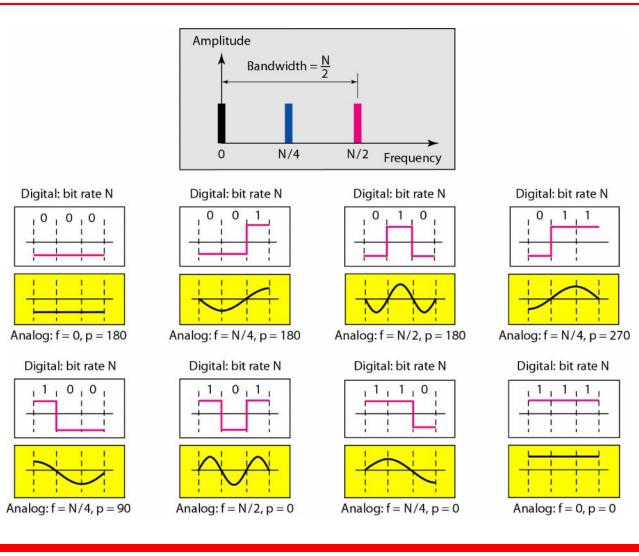
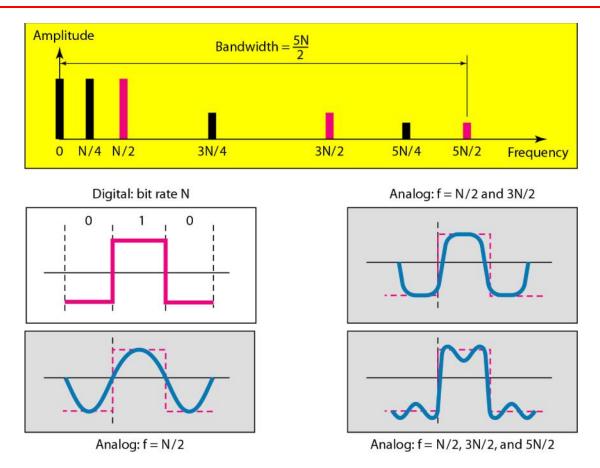


Figure 3.22 Simulating a digital signal with first three harmonics



Better Approximation: To make the shape of the analog signal look more like that of a digital signal, we need to add <u>more harmonics</u> of the frequencies. We need to increase the bandwidth.



In baseband transmission, the required bandwidth is proportional to the bit rate; if we need to send bits faster, we need more bandwidth.

Bit Rate	Harmonic 1	Harmonics 1, 3	Harmonics 1, 3, 5
n = 1 kbps	B = 500 Hz	B = 1.5 kHz	B = 2.5 kHz
n = 10 kbps	B = 5 kHz	B = 15 kHz	B = 25 kHz
n = 100 kbps	B = 50 kHz	B = 150 kHz	B = 250 kHz

 Table 3.2
 Bandwidth requirements

What is the required bandwidth of a low-pass channel if we need to send 1 Mbps by using baseband transmission?

Solution

The answer depends on the accuracy desired. a. The minimum bandwidth, is B = bit rate /2, or 500 kHz.

- **b.** A better solution is to use the first and the third harmonics with $B = 3 \times 500 \text{ kHz} = 1.5 \text{ MHz}$.
- c. Still a better solution is to use the first, third, and fifth harmonics with B = 5 × 500 kHz = 2.5 MHz.
 3.69

We have a low-pass channel with bandwidth 100 kHz. What is the maximum bit rate of this channel?

Solution

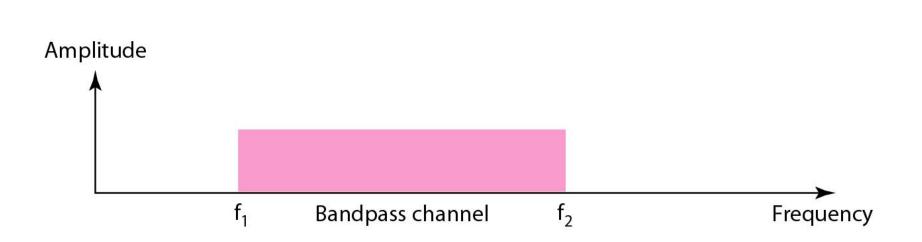
The maximum bit rate can be achieved if we use the first harmonic. The bit rate is 2 times the available bandwidth, or 200 kbps.

Bit rate = 2B

Broadband Transmission (Using Modulation)

- Broadband transmission or modulation means changing the digital signal to an analog signal for transmission.
- Modulation allows us to use a bandpass channel a channel with a bandwidth that does not start from zero.
- This type of channel is more available than a low-pass channel.

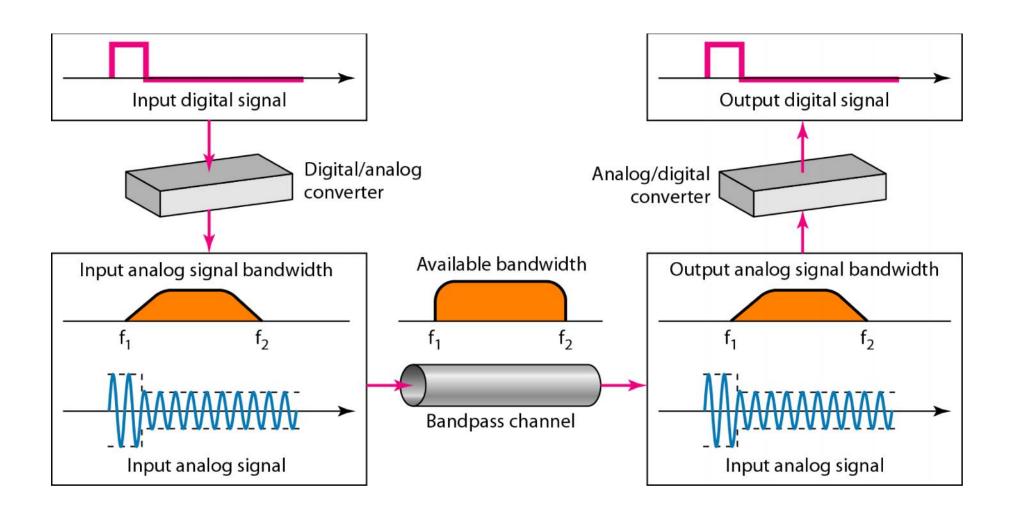
Figure 3.23 Bandwidth of a bandpass channel





If the available channel is a bandpass channel, we cannot send the digital signal directly to the channel; we need to convert the digital signal to an analog signal before transmission.

Figure 3.24 Modulation of a digital signal for transmission on a bandpass channel



Example 3.24

example of broadband transmission using An modulation is the sending of computer data through a telephone subscriber line, the line connecting a resident to the central telephone office. These lines are designed to carry voice with a limited bandwidth. The channel is considered a bandpass channel. We convert the digital signal from the computer to an analog signal, and send the analog signal. We can install two converters to change the digital signal to analog and vice versa at the receiving end. The converter, in this case, is called a *modem* which we discuss in detail in Chapter 5.

A second example is the digital cellular telephone. For better reception, digital cellular phones convert the analog voice signal to a digital signal (see Chapter 16). Although the bandwidth allocated to a company providing digital cellular phone service is very wide, we still cannot send the digital signal without conversion. The reason is that we only have a bandpass channel available between caller and callee. We need to convert the digitized voice to a composite analog signal before sending.

Signals travel through transmission media, which are not perfect. The imperfection causes signal impairment. This means that the signal at the beginning of the medium is not the same as the signal at the end of the medium. What is sent is not what is received. Three causes of impairment are attenuation, distortion, and noise.

Topics discussed in this section: Attenuation Distortion Noise Figure 3.25 Causes of impairment

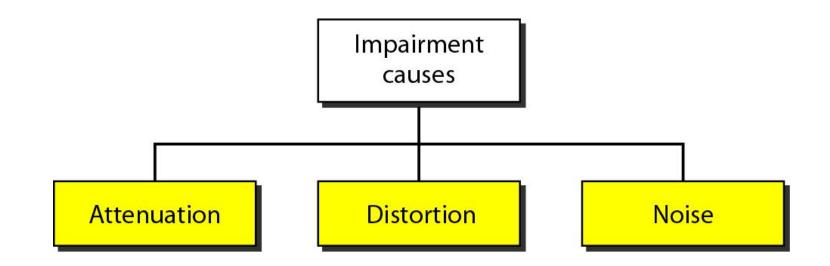
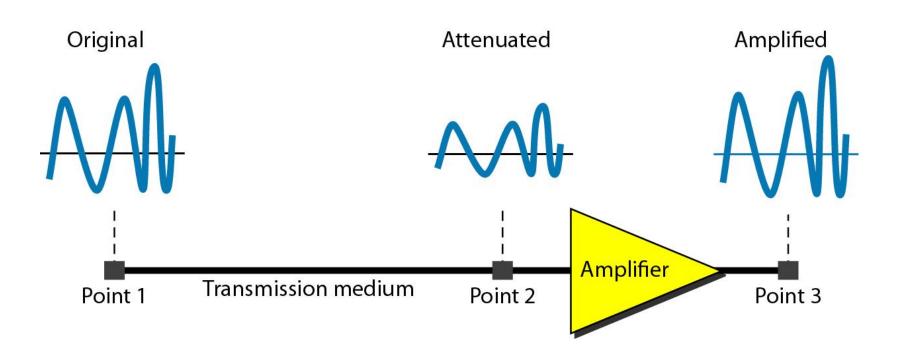


Figure 3.26 Attenuation



- Attenuation means a loss of energy.
- When a signal, simple or composite, travels through a medium, it loses some of its energy in overcoming the resistance of the medium.

Suppose a signal travels through a transmission medium and its power is reduced to one-half. This means that P_2 is $(1/2)P_1$. In this case, the attenuation (loss of power) can be calculated as

$$10 \log_{10} \frac{P_2}{P_1} = 10 \log_{10} \frac{0.5P_1}{P_1} = 10 \log_{10} 0.5 = 10(-0.3) = -3 \text{ dB}$$

A loss of 3 dB (-3 dB) is equivalent to losing one-half the power.

Note that the decibel is negative if a signal is attenuated and positive if a signal is amplified.

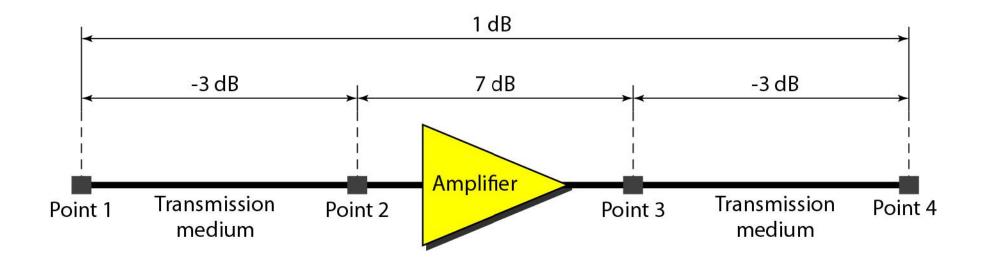
A signal travels through an amplifier, and its power is increased 10 times. This means that $P_2 = 10P_1$. In this case, the amplification (gain of power) can be calculated as

$$10 \log_{10} \frac{P_2}{P_1} = 10 \log_{10} \frac{10P_1}{P_1}$$

$$= 10 \log_{10} 10 = 10(1) = 10 \text{ dB}$$

One reason that engineers use the decibel to measure the changes in the strength of a signal is that decibel numbers can be added (or subtracted) when we are measuring several points (cascading) instead of just two. In Figure 3.27 a signal travels from point 1 to point 4. In this case, the decibel value can be calculated as

$$dB = -3 + 7 - 3 = +1$$



Sometimes the decibel is used to measure signal power in milliwatts. In this case, it is referred to as dB_m and is calculated as $dB_m = 10 \log 10 P_m$, where P_m is the power in milliwatts. Calculate the power of a signal with $dB_m = -30$.

Solution

We can calculate the power in the signal as

$$dB_{m} = 10 \log_{10} P_{m} = -30$$
$$\log_{10} P_{m} = -3 \qquad P_{m} = 10^{-3} \text{ mW}$$

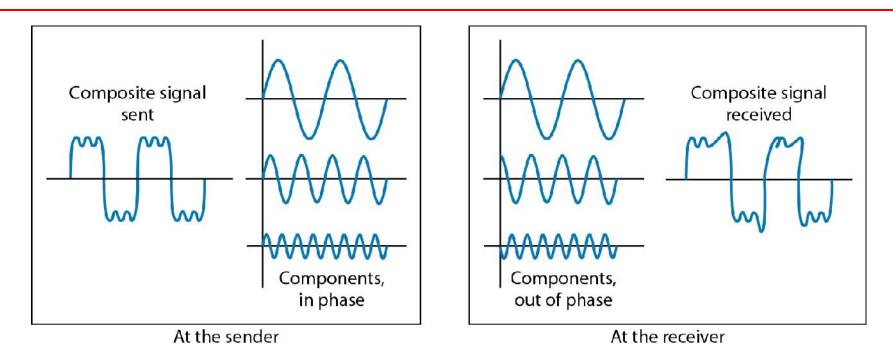
The loss in a cable is usually defined in decibels per kilometer (dB/km). If the signal at the beginning of a cable with -0.3 dB/km has a power of 2 mW, what is the power of the signal at 5 km? Solution

The loss in the cable in decibels is $5 \times (-0.3) = -1.5 \, dB$. We can calculate the power as

dB = 10 log₁₀
$$\frac{P_2}{P_1} = -1.5$$

 $\frac{P_2}{P_1} = 10^{-0.15} = 0.71$
 $P_2 = 0.71P_1 = 0.7 \times 2 = 1.4 \text{ mW}$

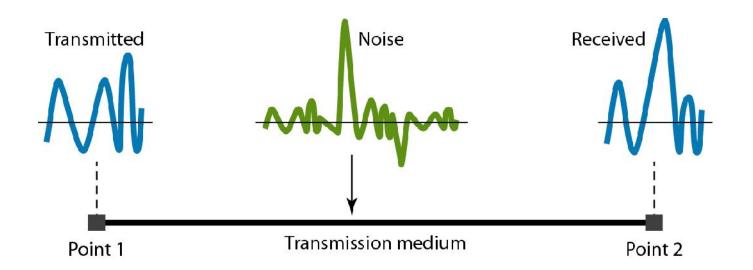
Figure 3.28 Distortion



- **Distortion** means that the signal changes its form or shape.
- Distortion can occur in a composite signal made of different frequencies.
- Each signal component has its own propagation speed through a medium and, therefore, its own delay in arriving at the final destination.

Distortion

- Attenuation distortion signal at different freq. attenuated at diff rate.
- Delay distortion signal at diff freq. delay at diff rate.
 - results in intersymbol interference



- Several types of noise, such as thermal noise, induced noise, crosstalk, and impulse noise, may corrupt the signal.
- <u>Thermal noise is the random motion of electrons in a wire which creates an extra signal not originally sent by the transmitter.</u>
- <u>Induced noise comes from sources such as motors and appliances.</u> These devices act as a sending antenna, and the transmission medium acts as the receiving antenna.
- <u>Crosstalk</u> is the effect of one wire on the other. One wire acts as a sending antenna and the other as the receiving antenna.
- <u>Impulse noise is a spike (a signal with high energy in a very short time) that comes from power lines, lightning, and so on.</u>

The power of a signal is 10 mW and the power of the noise is $1 \mu W$; what are the values of SNR and SNR_{dB}?

Solution The values of SNR and SNR_{dB} can be calculated as follows:

$$SNR = \frac{10,000 \ \mu W}{1 \ m W} = 10,000$$
$$SNR_{dB} = 10 \ \log_{10} 10,000 = 10 \ \log_{10} 10^4 = 40$$

Types of noise

- Thermal noise
- Intermodulation noise
- Crosstalk
- Impulse noise

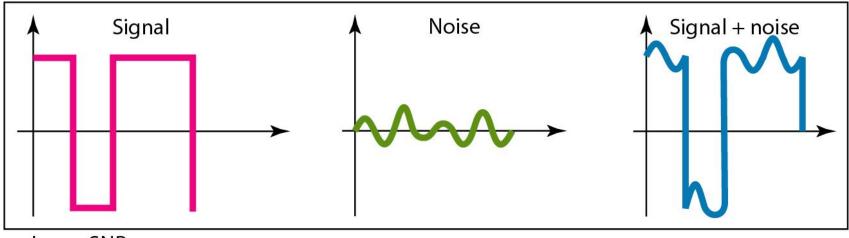


The values of SNR and SNR_{dB} for a <u>noiseless</u> channel are

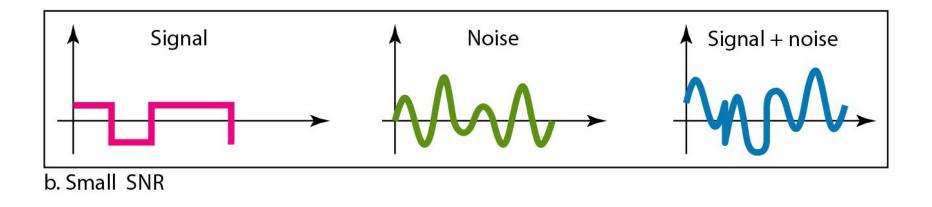
$$SNR = \frac{signal power}{0} = \infty$$
$$SNR_{dB} = 10 \log_{10} \infty = \infty$$

We can never achieve this ratio in real life; it is an ideal.

Figure 3.30 Two cases of SNR: a high SNR and a low SNR



a. Large SNR



A very important consideration in data communications is how fast we can send data, in bits per second, over a channel. Data rate depends on <u>three factors</u>:

- **1**. The bandwidth available
- 2. The level of the signals we use
- **3**. The quality of the channel (the level of noise)

Topics discussed in this section:

Noiseless Channel: Nyquist Bit Rate Noisy Channel: Shannon Capacity Using Both Limits

Noiseless Channel: Nyquist Bit Rate

BitRate = $2 \times \text{bandwidth} \times \log_2 L$



Increasing the levels of a signal may reduce the reliability of the system.

Does the **Nyquist theorem** bit rate agree with the intuitive bit rate described in baseband transmission?

Solution

They match when we have only <u>two levels</u>. We said, in baseband transmission, the bit rate is 2 times the bandwidth if we use only the first harmonic in the worst case. However, the Nyquist formula is more general than what we derived intuitively; it can be applied to baseband transmission and modulation. Also, it can be applied when we have two or more levels of signals.



Consider a noiseless channel with a bandwidth of 3000 Hz transmitting a signal with two signal levels. The maximum bit rate can be calculated as

 $BitRate = 2 \times 3000 \times \log_2 2 = 6000 \text{ bps}$



Consider the same noiseless channel transmitting a signal with four signal levels (for each level, we send 2 bits). The maximum bit rate can be calculated as

BitRate = $2 \times 3000 \times \log_2 4 = 12,000$ bps

We need to send 265 kbps over a noiseless channel with a bandwidth of 20 kHz. How many signal levels do we need?

Solution

We can use the Nyquist formula as shown:

 $265,000 = 2 \times 20,000 \times \log_2 L$ $\log_2 L = 6.625$ $L = 2^{6.625} = 98.7$ levels

Since this result is not a power of 2, we need to either increase the number of levels or reduce the bit rate. If we have 128 levels, the bit rate is 280 kbps. If we have 64 levels, the bit rate is 240 kbps.

In reality, we cannot have a noiseless channel; the channel is always noisy. In 1944, Claude Shannon introduced a formula, called the Shannon capacity, to determine the theoretical highest data rate for a noisy channel:

$Capacity = bandwidth \times log_2 (1 + SNR)$

Note that in the Shannon formula there is no indication of the signal level, which means that no matter how many levels we have, we cannot achieve a data rate higher than the capacity of the channel. In other words, the formula defines a characteristic of the channel, not the method of transmission.

Consider an extremely noisy channel in which the value of the signal-to-noise ratio is almost zero. In other words, the noise is so strong that the signal is faint. For this channel the capacity C is calculated as

 $C = B \log_2 (1 + SNR) = B \log_2 (1 + 0) = B \log_2 1 = B \times 0 = 0$

This means that the capacity of this channel is zero regardless of the bandwidth. In other words, we cannot receive any data through this channel.

We can calculate the theoretical highest bit rate of a regular telephone line. A telephone line normally has a bandwidth of 3000. The signal-to-noise ratio is usually 3162. For this channel the capacity is calculated as

 $C = B \log_2 (1 + \text{SNR}) = 3000 \log_2 (1 + 3162) = 3000 \log_2 3163$ $= 3000 \times 11.62 = 34,860 \text{ bps}$

This means that the highest bit rate for a telephone line is 34.860 bps. If we want to send data faster than this, we can either increase the bandwidth of the line or improve the signal-to-noise ratio.

Example 3.39

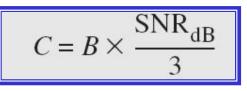
The signal-to-noise ratio is often given in decibels. Assume that $SNR_{dB} = 36$ and the channel bandwidth is 2 MHz. The theoretical channel capacity can be calculated as

$$SNR_{dB} = 10 \log_{10} SNR \implies SNR = 10^{SNR_{dB}/10} \implies SNR = 10^{3.6} = 3981$$

 $C = B \log_2 (1 + SNR) = 2 \times 10^6 \times \log_2 3982 = 24 \text{ Mbps}$



For practical purposes, when the SNR is very high, we can assume that SNR + 1 is almost the same as SNR. In these cases, the theoretical channel capacity can be simplified to



For example, we can calculate the theoretical capacity of the previous example as

$$C = 2 \text{ MHz} \times \frac{36}{3} = 24 \text{ Mbps}$$

We have a channel with a 1-MHz bandwidth. The SNR for this channel is 63. What are the appropriate bit rate and signal level?

Solution

First, we use the Shannon formula to find the upper limit.

$$C = B \log_2 (1 + \text{SNR}) = 10^6 \log_2 (1 + 63) = 10^6 \log_2 64 = 6 \text{ Mbps}$$

Example 3.41 (continued)

The Shannon formula gives us 6 Mbps, the upper limit. For better performance we choose something lower, 4 Mbps, for example. Then we use the Nyquist formula to find the number of signal levels.

$$4 \text{ Mbps} = 2 \times 1 \text{ MHz} \times \log_2 L \implies L = 4$$



The Shannon capacity gives us the upper limit; the Nyquist formula tells us how many signal levels we need.

One important issue in networking is the performance of the network - how good is it? We discuss quality of service, an overall measurement of network performance, in greater detail in Chapter 24. In this section, we introduce terms that we need for future chapters.

Topics discussed in this section:

Bandwidth Throughput Latency (Delay) Bandwidth-Delay Product

3.107



In networking, we use the term bandwidth in two contexts.

- The first, bandwidth in hertz (Hz), refers to the <u>range of frequencies</u> in a composite signal or the range of frequencies that a channel can pass.
- The second, bandwidth in bits per second (bps), refers to the <u>speed of bit transmission</u> in a channel or link.

Example 3.42

The bandwidth of a subscriber line is 4 kHz for voice or data. The bandwidth of this line for data transmission can be up to 56,000 bps using a sophisticated modem to change the digital signal to analog.

Example 3.43

If the telephone company improves the quality of the line and increases the bandwidth to 8 kHz, we can send 112,000 bps by using the same technology as mentioned in Example 3.42.

Throughput

- The throughput is a measure of <u>how fast</u> we can actually send data through a network.
- Although, at first glance, bandwidth in bits per second and throughput seem the same, they are different.
- A link may have a bandwidth of B bps, but we can only send T bps through this link with T < B, always.
- The bandwidth is a potential measurement of a link; the throughput is an actual measurement of how fast we can send data.
- For example:

We may have a link with a bandwidth of 1 Mbps, but the devices connected to the end of the link may handle only 200 kbps. This means that we cannot send more than 200 kbps through this link.

Example 3.44

A network with bandwidth of 10 Mbps can pass only an average of 12,000 frames per minute with each frame carrying an average of 10,000 bits. What is the throughput of this network?

Solution

We can calculate the throughput as

Throughput = $\frac{12,000 \times 10,000}{60}$ = 2 Mbps

The throughput is almost <u>one-fifth</u> of the bandwidth in this case.

Latency (Delay)

- Defines how long it takes for an **entire message to completely arrive** at the destination from the time the first bit is sent out from the source.
- Made of four components: propagation time, transmission time, queuing time and processing delay.

Latency = propagation time + transmission time + queuing time + processing delay

Propagation Time

 $Propagation time = \frac{Distance}{Propagation speed}$



What is the propagation time if the distance between the two points is 12,000 km? Assume the propagation speed to be 2.4×10^8 m/s in cable.

Solution

We can calculate the propagation time as

 $Propagation time = \frac{Distance}{Propagation speed}$

Propagation time =
$$\frac{12,000 \times 1000}{2.4 \times 10^8} = 50 \text{ ms}$$

The example shows that a bit can go over the Atlantic Ocean in only 50 ms if there is a direct cable between the source and the destination.

Example 3.46

What are the propagation time and the transmission time for a 2.5-kbyte message (an e-mail) if the bandwidth of the network is 1 Gbps? Assume that the distance between the sender and the receiver is 12,000 km and that light travels at 2.4×10^8 m/s.

Solution

We can calculate the propagation and transmission time as shown on the next slide:

Example 3.46 (continued)Transmission time =
$$\frac{Message size}{Bandwidth}$$
Propagation time = $\frac{Distance}{Propagation speed}$ Propagation time = $\frac{12,000 \times 1000}{2.4 \times 10^8} = 50 \text{ ms}$ Transmission time = $\frac{2500 \times 8}{10^9} = 0.020 \text{ ms}$

Note that in this case, because the message is short and the bandwidth is high, the dominant factor is the propagation time, not the transmission time. The transmission time can be ignored.

Example 3.47

What are the propagation time and the transmission time for a 5-Mbyte message (an image) if the bandwidth of the network is 1 Mbps? Assume that the distance between the sender and the receiver is 12,000 km and that light travels at 2.4×10^8 m/s.

Solution

We can calculate the propagation and transmission times as shown on the next slide.



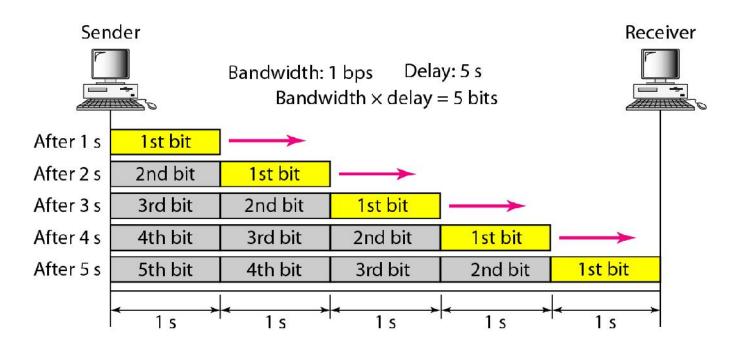
Propagation time =
$$\frac{12,000 \times 1000}{2.4 \times 10^8} = 50 \text{ ms}$$

Transmission time =
$$\frac{5,000,000 \times 8}{10^6} = 40 \text{ s}$$

Note that in this case, because the message is very long and the bandwidth is not very high, the dominant factor is the transmission time, not the propagation time. The propagation time can be ignored.

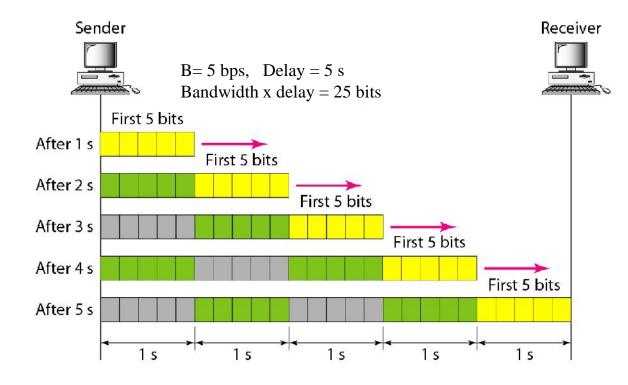
Bandwidth-Delay Product

Figure 3.31 Filling the link with bits for <u>case 1</u>



- The product 1 x 5 is the maximum number of bits that can fill the link.
- There can be no more than 5 bits at any time on the link.

Figure 3.32 Filling the link with bits in <u>case 2</u>



There can be maximum $5 \times 5 = 25$ bits on the line. The reason is that, at each second, there are 5 bits on the line; the duration of each bit is 0.2 s.

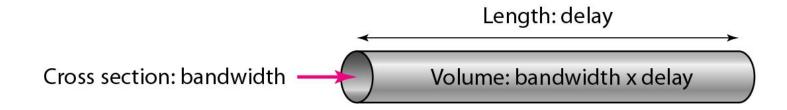




The bandwidth-delay product defines the number of bits that can fill the link.

Example 3.48

We can think about the link between two points as a pipe. The cross section of the pipe represents the bandwidth, and the length of the pipe represents the delay. We can say the volume of the pipe defines the bandwidth-delay product, as shown in Figure 3.33. **Figure 3.33** Concept of bandwidth-delay product



Jitter

- Another performance issue that is related to delay is jitter.
- We can roughly say that jitter is a problem if different packets of data encounter <u>different delays</u> and the application using the data at the receiver site is time-sensitive (audio and video data, for example).
- If the delay for the first packet is 20 ms, for the second is 45 ms, and for the third is 40 ms, then the real-time application that uses the packets endures jitter.