

---

# *Redes de Computadores*

## **The Physical Layer**

*Manuel P. Ricardo*

*Faculdade de Engenharia da Universidade do Porto*

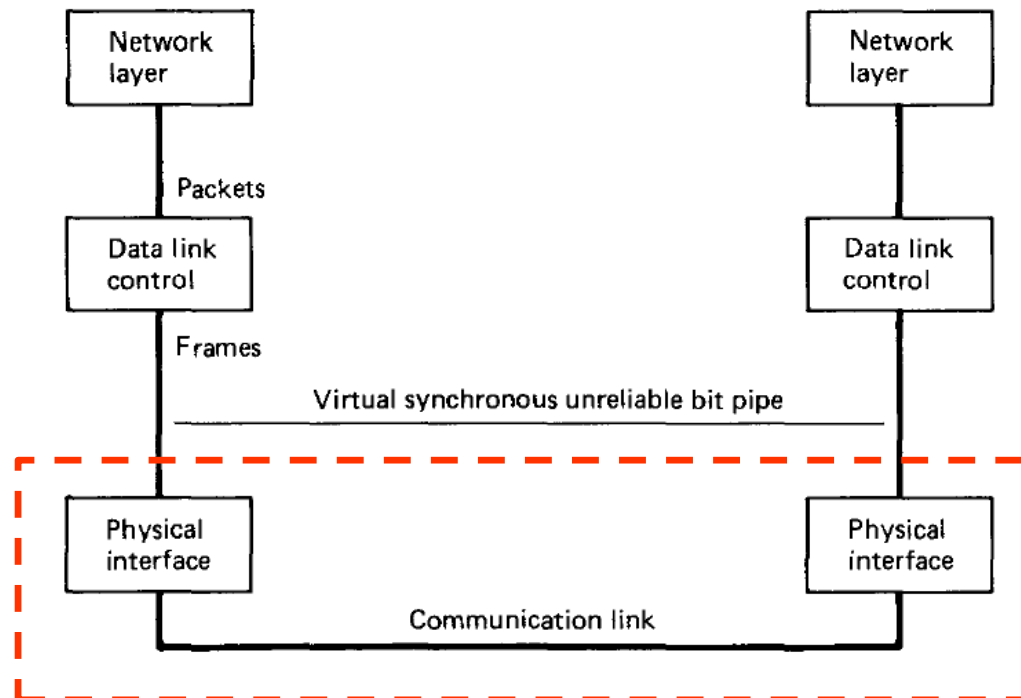
- 
- » *What service does the Physical Layer offer to Data Link Layer?*
  - » *How to encode a sequence of bits into an analogue signal?*
  - » *Why does the received signal  $r(t)$  differ from the transmitted signal  $s(t)$ ?*
  - » *What is the difference between baudrate and bitrate?*
  - » *What are the advantages of the Manchester code over the NRZ code?*
  - » *What are the common digital modulations?*
  - » *What is the maximum capacity of a communications channel?*
  - » *What types of media exist and what are their main characteristics?*
  - » *What is dB, dBW, dBm, Gain, and Attenuation?*
  - » *How does the attenuation of a wireless channel vary with distance and wavelength?*

# Service Provided by the Physical Layer

Physical layer *→ aqui vai estar o canal rádio/cabo*  
*→ canal real que serve para fazer comunicação de um lado para o outro*

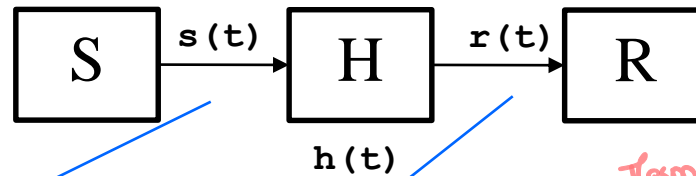
- » real communication channels used by the network
- » interfaces required to transmit and receive digital data
- » appears to higher layers as an unreliable virtual bit pipe

► Para o canal superior  
vai ser um tubo que  
pode perder informa-  
ção



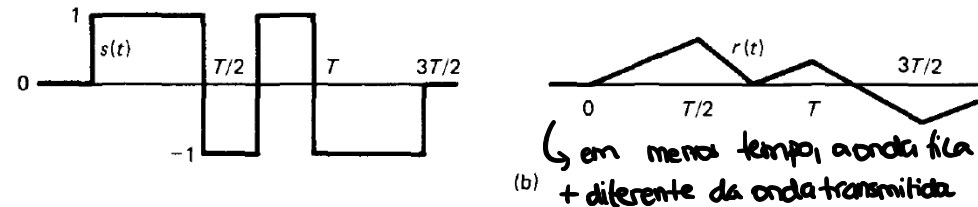
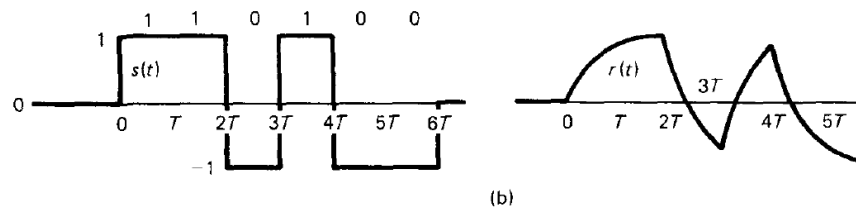
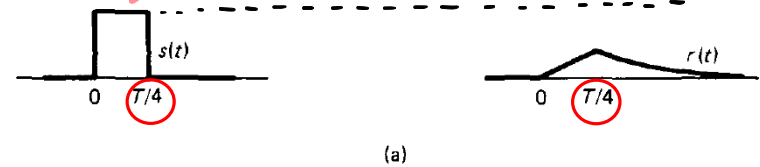
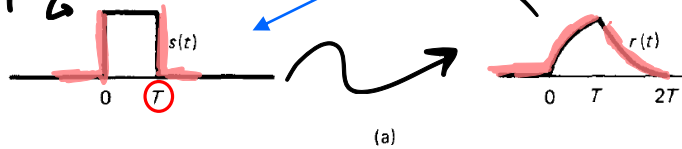
# Transmission Channel Modifies Input Signal $s(t)$

Vai levar algum tempo a mudar de níveis.  
Não é instantâneo  
como aqui?



Tempo sistema  
Transição + rápida

O canal pode não ter tempo para atingir o nível desejado

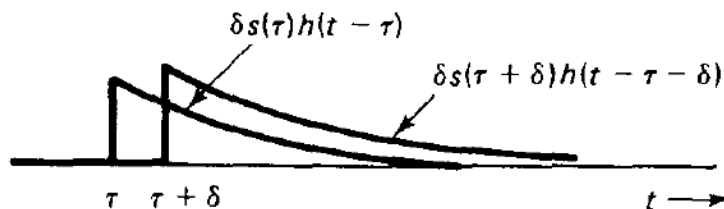
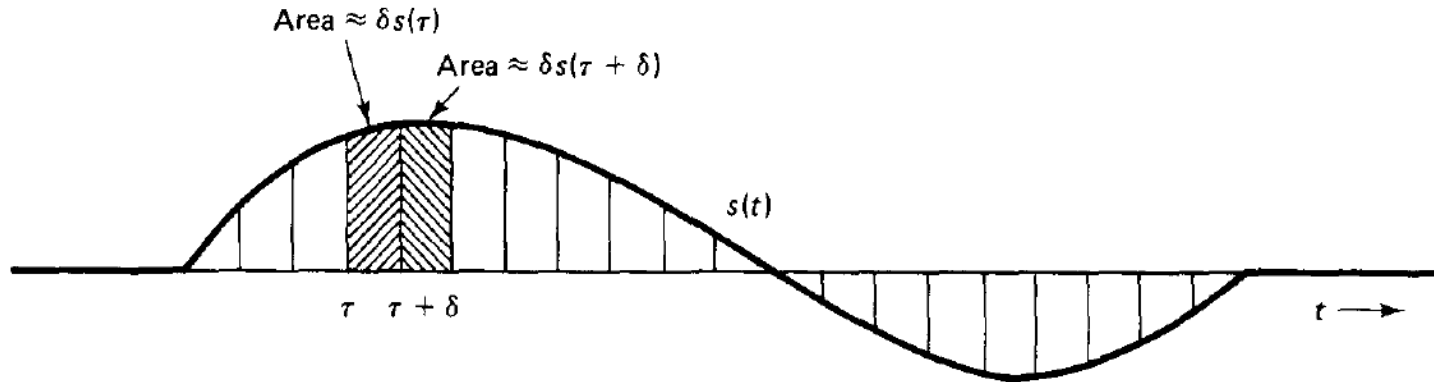


**Figure 2.3** Relation of input and output waveforms for a communication channel with filtering. Part (a) shows the response  $r(t)$  to an input  $s(t)$  consisting of a rectangular pulse, and part (b) shows the response to a sequence of pulses. Part (b) also illustrates the NRZ code in which a sequence of binary inputs (1 1 0 1 0 0) is mapped into rectangular pulses. The duration of each pulse is equal to the time between binary inputs.

**Figure 2.4** Relation of input and output waveforms for the same channel as in Fig. 2.3. Here the binary digits enter at 4 times the rate of Fig. 2.3, and the rectangular pulses last one-fourth as long. Note that the output  $r(t)$  is more distorted and more attenuated than that in Fig. 2.3.

Há sistemas que reagem mais rapidamente

$x(t)$  is the convolution of  $s(t)$  and  $h(t)$



Impulso que varia, instantaneamente colocado no canal  $h(t)$

$$r(t) = \int_{-\infty}^{+\infty} s(\tau)h(t - \tau)d\tau$$

Signal recebido é a convulsão do sinal enviado

# The Fourier Transform

- Using Fourier transforms

*sinais enviados*  $\rightarrow$   $S(f) = \int_{-\infty}^{+\infty} s(\tau) e^{-j2\pi f\tau} d\tau$

$t \rightarrow f$

*Canal de transmissão*  $\rightarrow$   $H(f) = \int_{-\infty}^{+\infty} h(\tau) e^{-j2\pi f\tau} d\tau$

and, in the frequency domain

$R(f) = S(f) \times H(f)$

*Resposta e frequência do canal de transmissão*

*sinai recebido*

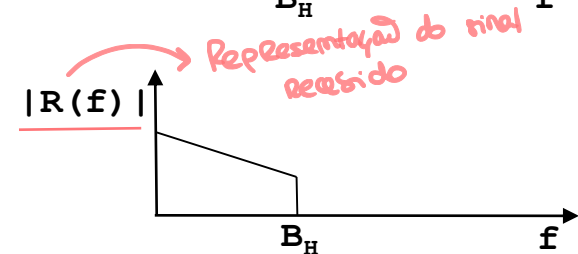
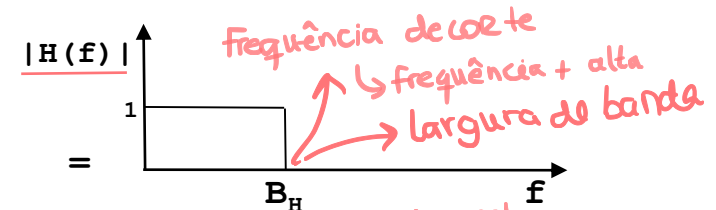
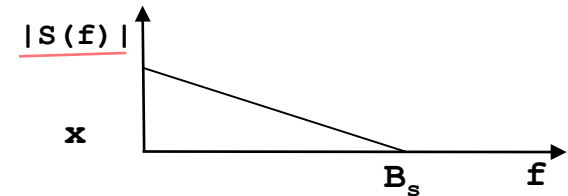
$\hookrightarrow$  *sinai enviado*

- Thus,  $\mathbf{r}(t)$  depends on  $\mathbf{B}_H$

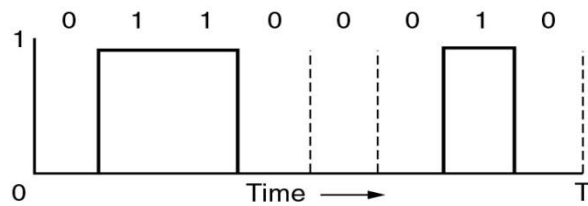
$r(t) = \int_{-\infty}^{+\infty} R(f) e^{j2\pi ft} df = \int_{-B_H}^{+B_H} R(f) e^{j2\pi ft} df$

*transformada inversa*

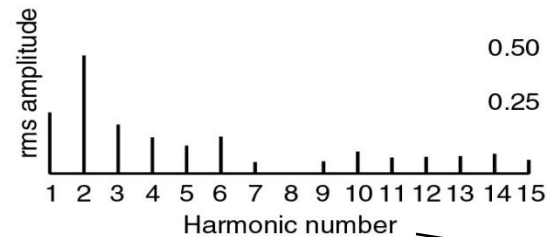
$f \rightarrow t$



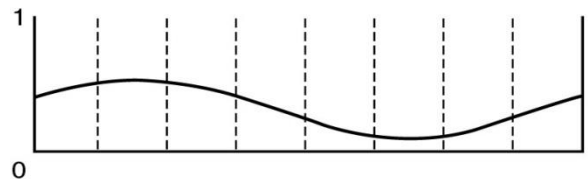
# Bandwidth-Limited Signals



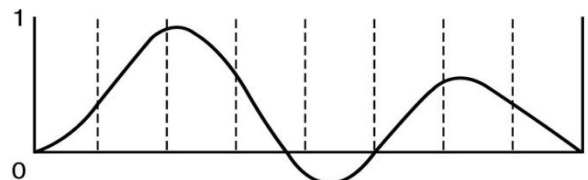
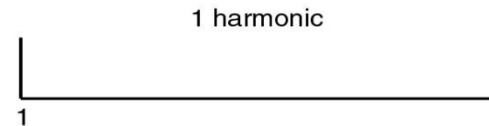
(a)



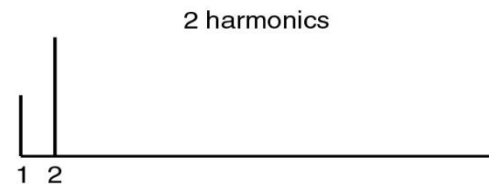
$$\frac{i}{T} \text{ Hz}$$



(b)



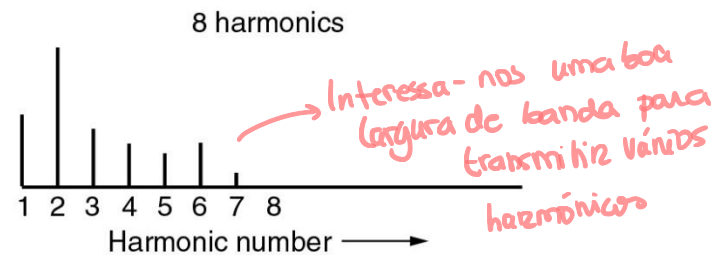
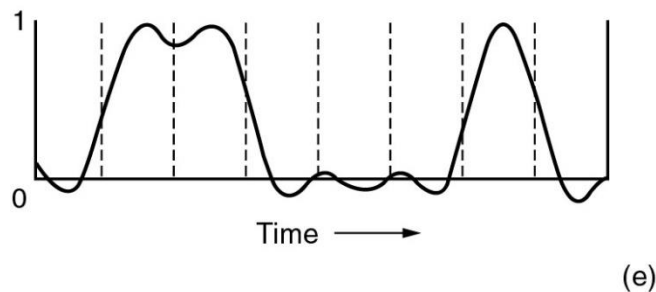
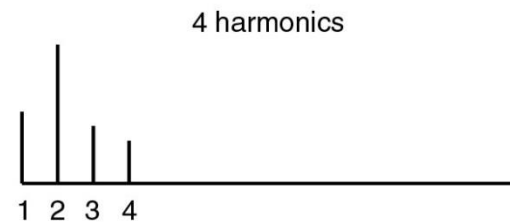
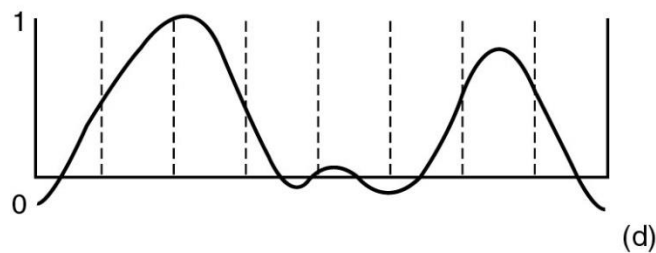
(c)



(a) binary signal and its root-mean-square Fourier amplitudes

(b) – (c) Successive approximations to the original signal

# Bandwidth-Limited Signals



*(d) – (e) Successive approximations to the original signal.*



# *Reconstructing the Signal the Receiver*

---

- ♦ In digital communications, the receiver
  - » samples  $r(t)$  in order to decide about the bits transmitted
  
- ♦ Nyquist showed that
  - » a signal  $r(t)$  having a bandwidth  $B$  Hz
  - » can be fully reconstructed
  - » if sampled at rate  $2B$  sample/s
  
  - » sampling at higher rate does not provide additional information

# Transmitting Information

◆ Let us assume

- » a square wave  $v(t)$  alternating between  $-5\text{ V}$  (bit 0) and  $5\text{ V}$  (bit 1),
- » passing through a lowpass channel  $B_H = 3\text{ kHz}$

Onda de tensão que varia entre

2 níveis

freq. de corte

signal recebido não pode ter frequência superior a essa

◆ If receiver samples the signal at  $2B_H = 6\text{ ksample/s}$ , it receives

- » a bitrate of  $C = 2 \cdot B_H = 6\text{ kbit/s}$  (1 sample - 1 bit of information)

↳ Aproximação inicial do débito

Por cada amostra quantos bits são transmitidos

◆ However,

- » if  $M=4$  levels are used to encode information

–  $-5\text{V}(00)$ ,  $-2\text{V}(01)$ ,  $2\text{V}(10)$ ,  $5\text{V}(11)$

$$C = 2B \log_2(M)$$

↳ A cada nível transmitem-se 2 bits

- » Then, the channel capacity becomes  $C = 2B \log_2(M) = 2 \times 3k \times 2 = 12\text{ kbit/s}$

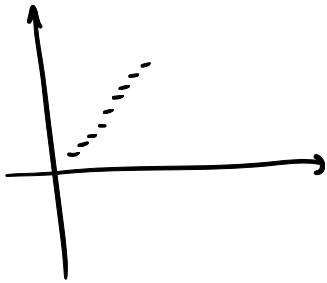
↳ 4 canais → 2 bits por sinal

- »  **$2B$  expresses the channel baudrate in symbol/s or baud**

## *To Think*

---

- ♦ Can we transmit an infinite number of bit/s in a channel of bandwidth  $B = 3$  kHz by increasing the number of levels  $M$ ?



$$C = 2B \log_2(M)$$

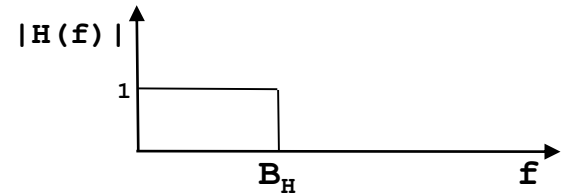
→ Não é possível, pois se tivermos um número muito grande de níveis, quando fazemos a transmissão para o receptor, pode haver ruído (sinal um pouco + a cima de + baixo) e como com muitos canais, os níveis são muito próximos, o sinal pode ser mal interpretado → Probabilidade de erro aumenta

# *Baseband / Passband Transmission*

---

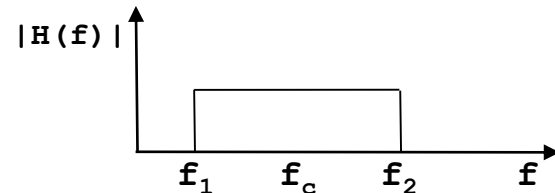
## ♦ Baseband transmission

- » signal has frequencies from zero up to a maximum  $B_H$
- » common for wires



## ♦ Passband transmission

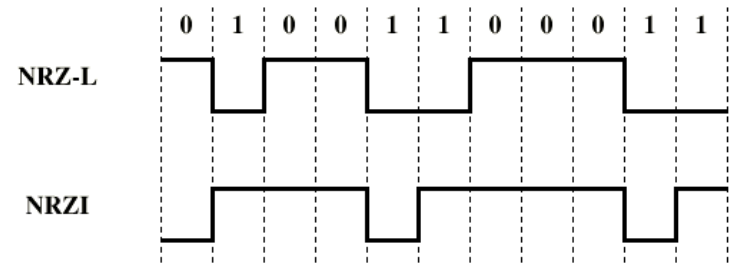
- » signal uses band of frequencies around the **frequency of the carrier  $f_c$**
- » common for wireless and optical channels



# Baseband Transmission - Common Codes

- ◆ NRZ-L (Non Return to Zero - Level)

- » Two levels representing 0 and 1

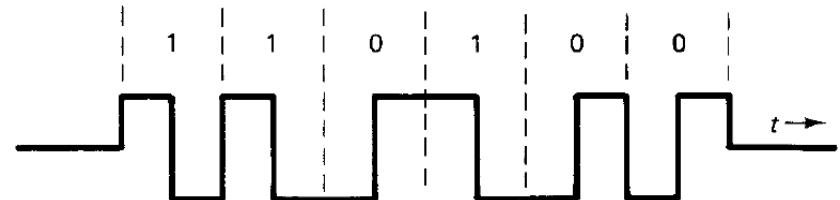


- ◆ NRZ-I (Non Return to Zero - Inverted)

- » Change of level represents a 1

- ◆ **Manchester**

- » Transition in the middle of the bit
- » 1: positive → negative
- » 0: negative → positive
- » Used in Ethernet (IEEE 802.3)



- ◆ There are many codes ...

# *Clock Recovery*

---

- ◆ To decode the symbols, signals need sufficient transitions
  - » Otherwise long runs of 0s (or 1s) are confusing, e.g.:



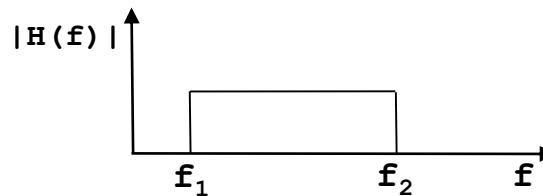
- ◆ Strategies:
  - » Manchester coding, mixes clock signal in every symbol
  - » 4B/5B maps 4 data bits to 5 coded bits with 1s and 0s:

Data	Code	Data	Code	Data	Code	Data	Code
0000	11110	0100	01010	1000	10010	1100	11010
0001	01001	0101	01011	1001	10011	1101	11011
0010	10100	0110	01110	1010	10110	1110	11100
0011	10101	0111	01111	1011	10111	1111	11101

# *Bandpass Transmission*

---

- ♦ Some physical channels are bandpass



- ♦ Technique used to enable  $\mathbf{s(t)}$  to pass through  $\mathbf{h(t)}$ 
  - » Modulation

## *To Think*

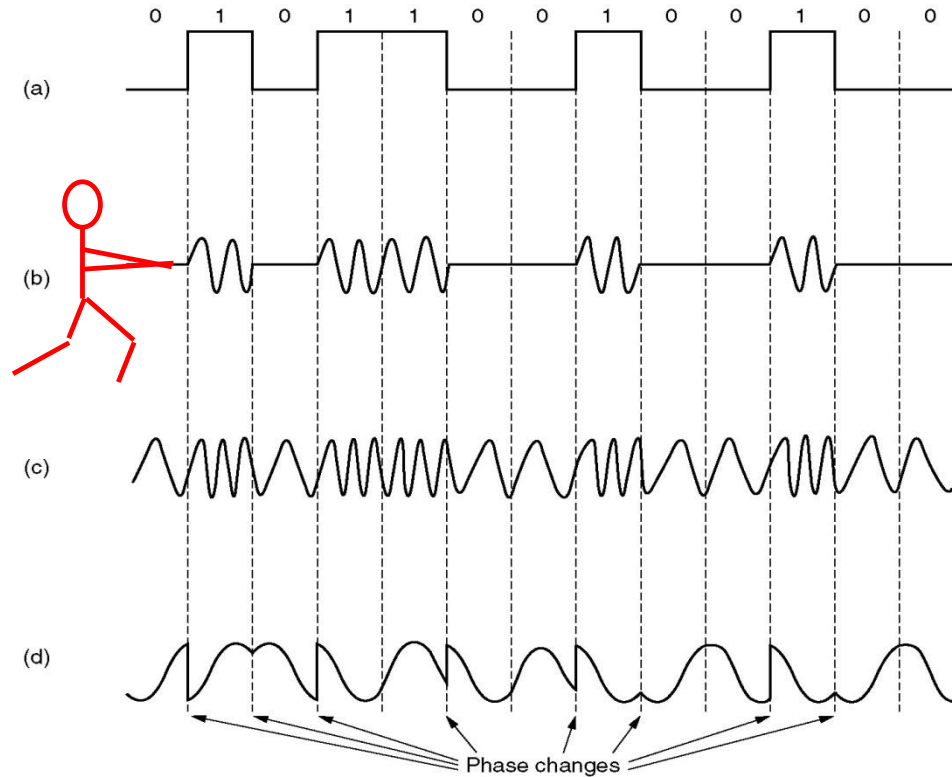
---

- ◆ How to transmit bits using a continuous carrier?



# *Types of Modulations*

---



(a) A binary signal

(b) Amplitude modulation

(c) Frequency modulation

(d) Phase modulation

# Amplitude and Phase Modulations

## ◆ Amplitude Modulation

information coded in the carrier's amplitude

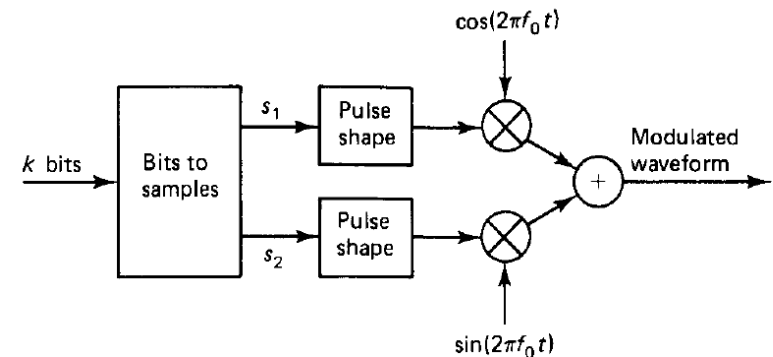
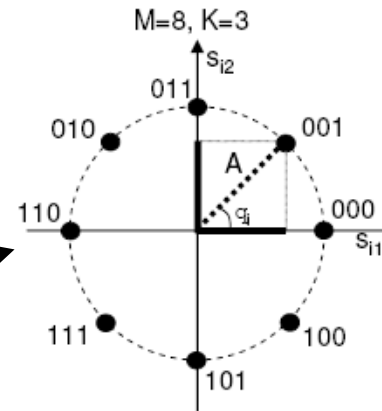
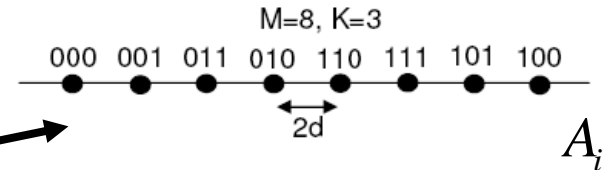
$$s(t) = A_i \cos(2\pi f_c t)$$

## ◆ Phase Modulation

information coded in carrier's phase

$$s(t) = A \cos(\theta_i + 2\pi f_c t)$$

$$\begin{aligned} s(t) &= A \cos(\theta_i + 2\pi f_c t) = \\ &= A \cos(\theta_i) \cos(2\pi f_c t) - A \sin(\theta_i) \sin(2\pi f_c t) = \\ &= s_1(t) \cos(2\pi f_c t) + s_2(t) \sin(2\pi f_c t) \end{aligned}$$



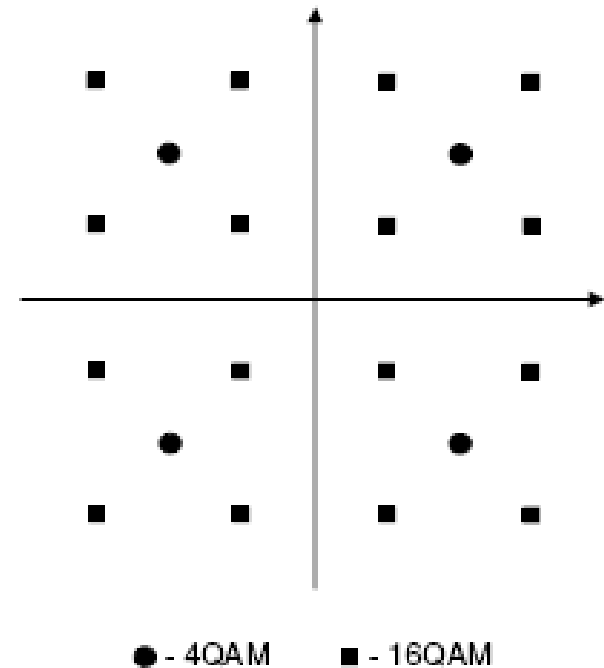
(a) Modulator

## ◆ $K = \log_2 M$ bits sent over a time symbol interval

# Quadrature Amplitude Modulation

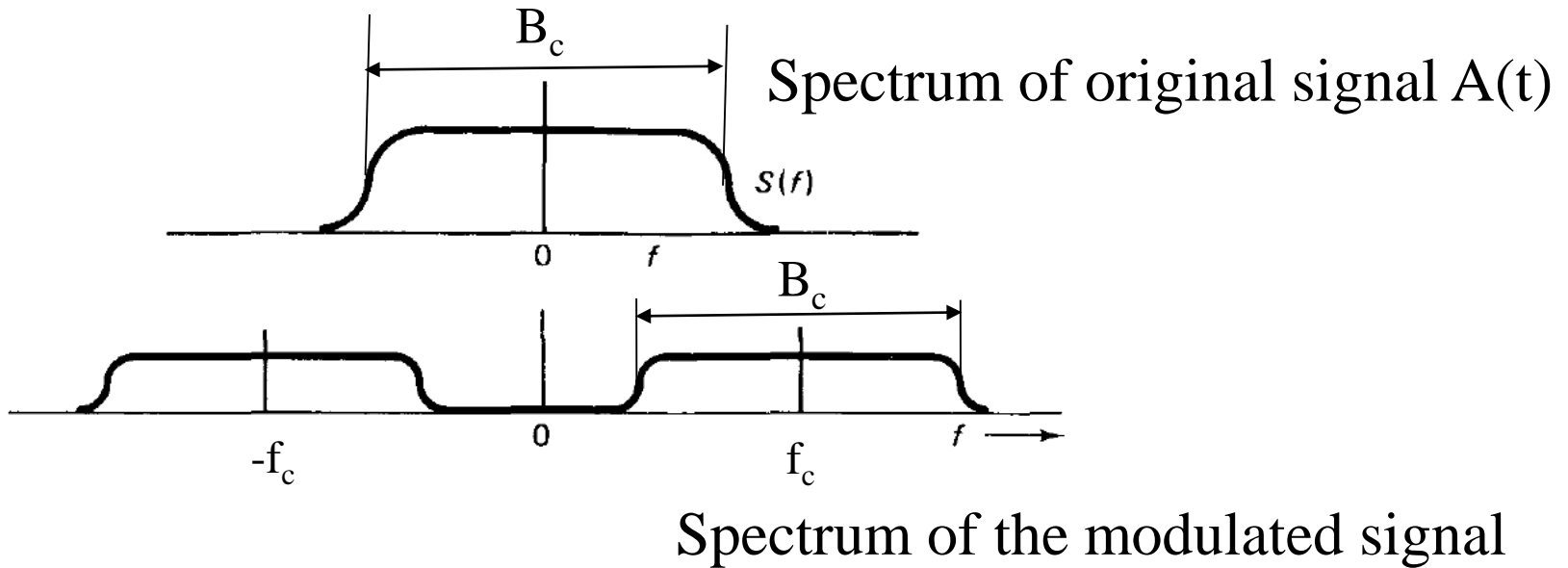
- ◆ Quadrature Amplitude Modulation (**M-QAM**)  
information coded both in amplitude and phase

$$s(t) = A_i \cos(\theta_i + 2\pi f_c t)$$



# *Amplitude Modulation - Representation in the Frequency domain*

---



# *Shannon's Law*

---

- ◆ Noise imposes the limit on the number levels  $M$  (bit/symbol)
  - » Noise high  $\rightarrow$  low  $M$
  - » or, high Signal to Noise Ratio (SNR)  $\rightarrow$  high  $M$
- ◆ Maximum theoretical capacity of a channel,  $C$  (bit/s)

$$C = B_c \log_2 \left( 1 + \frac{P_r}{N_0 B_c} \right)$$

- »  $B_c$  – bandwidth of the channel (Hz) (see last slide)  
 $B_c$  = sampling rate
- »  $P_r$  – signal power as seen by receiver (W)
- »  $N_0 B_c$  - noise power within the bandwidth  $B_c$ , as seen by receiver (W)
- »  $N_0$  – White noise; noise power per unit bandwidth (W/Hz)

## *Example*

---

- ♦ If a bandpass channel has a bandwidth  $B_c = 100 \text{ kHz}$  and Signal to Noise ratio (SNR) at the receiver is

- »  $P_r/(N_0 B_c) = 7 \rightarrow C = 100k \log_2(1 + 7) = 300k \text{ bit/s}$

- »  $P_r/(N_0 B_c) = 255 \rightarrow C = 100k \log_2(1 + 255) = 800k \text{ bit/s}$

- ♦ Power expressed in **W, dBW, or dBm**

- »  $P_{\text{dBW}} = 10 \log_{10} P$  :  $P = 100 \text{ mW} \rightarrow P_{\text{dBW}} = 10 \log_{10}(100 \cdot 10^{-3}) = -10 \text{ dBW}$

- »  $P_{\text{dBm}} = 10 \log_{10}(P/1 \text{ mW})$  :  $P = 100 \text{ mW} \rightarrow P_{\text{dBm}} = 10 \log_{10}(100) = 20 \text{ dBm}$

# *Guided Transmission*

---

- ♦ Twisted Pair
- ♦ Coaxial Cable
- ♦ Fiber Optics

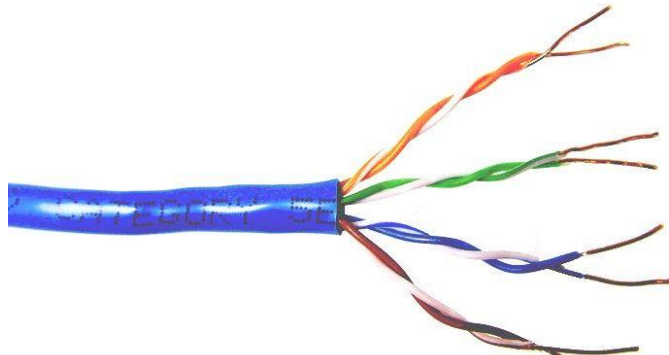
# *Guided Transmission*

---

- ◆ Coaxial cable



- ◆ Unshielded twisted pair



- ◆ Fiber optic





# *Twisted Pair*

---



(a)



(b)

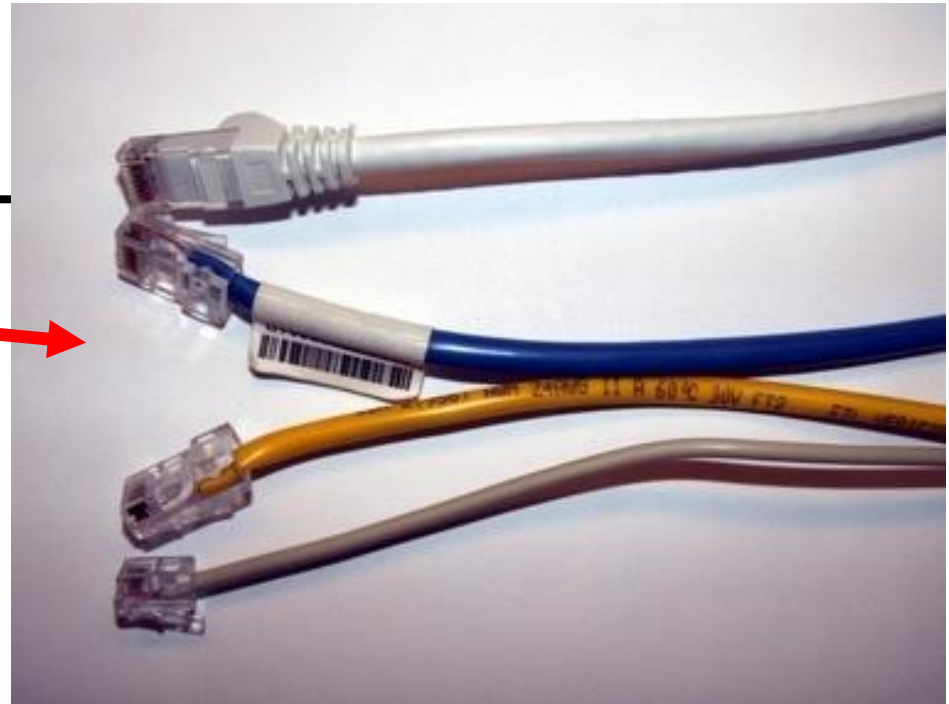
(a) Category 3 UTP.

(b) Category 5 UTP.

## *UTP Cables*

---

- ◆ From top to bottom:  
Cat. 6, Cat. 5e, Cat. 5, Cat. 3
- ◆ Cat 3, 16 MHz bandwidth
- ◆ Cat. 5 / 5e, 100MHz
- ◆ Cat. 6, 250MHz
- ◆ Cat. 6a, 500MHz
- ◆ Cat. 7, 600MHz
- ◆ Typical attenuations 2 – 25 dB/100 m



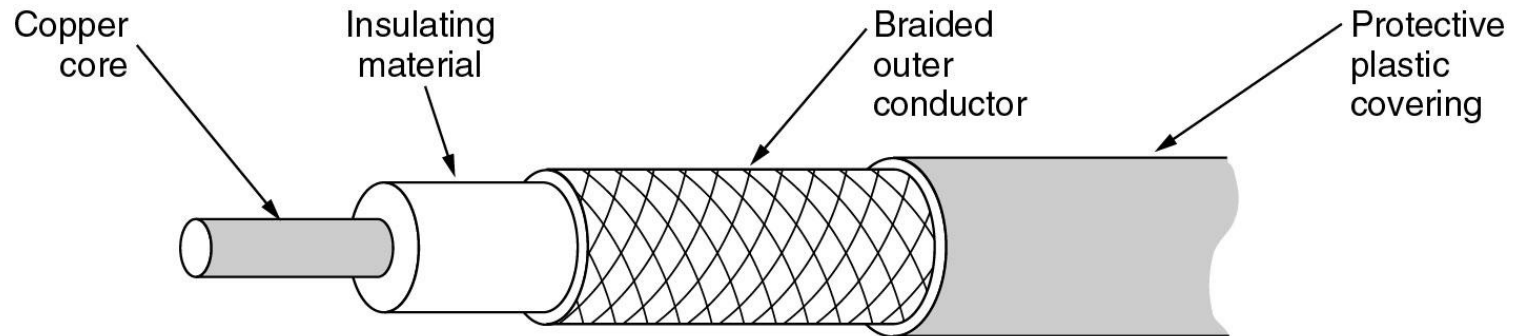
## *dB, dBm, Gain, Attenuation*

- ♦ Attenuation and Gain of the channel are related issues
- ♦ In Watts  $\rightarrow P_r = P_t * \text{Gain}$
- ♦ In dB,
  - »  $10\log_{10}(P_r) = 10\log_{10}(P_t * \text{Gain}) = 10\log_{10}(P_t) + 10\log_{10}(\text{Gain})$
  - »  $P_{r_{\text{dBW}}} = P_{t_{\text{dBW}}} + \text{Gain}_{\text{dB}}$     or     $P_{r_{\text{dBm}}} = P_{t_{\text{dBm}}} + \text{Gain}_{\text{dB}}$
  - » If  $\text{Gain} = 0.01$  and  $P_{t_{\text{dBm}}} = 30 \text{ dBm}$  (1W)
    - $\text{Gain}_{\text{dB}} = 10\log_{10}(0.01) = -20\text{dB}$
    - $P_{r_{\text{dBm}}} = P_{t_{\text{dBm}}} + \text{Gain}_{\text{dB}} = 30 - 20 = 10\text{dBm} = 10\text{mW}$
- ♦  $\text{Gain} = -20\text{dB} \leftrightarrow \text{Attenuation} = 20\text{dB}$

# *Coaxial Cable*

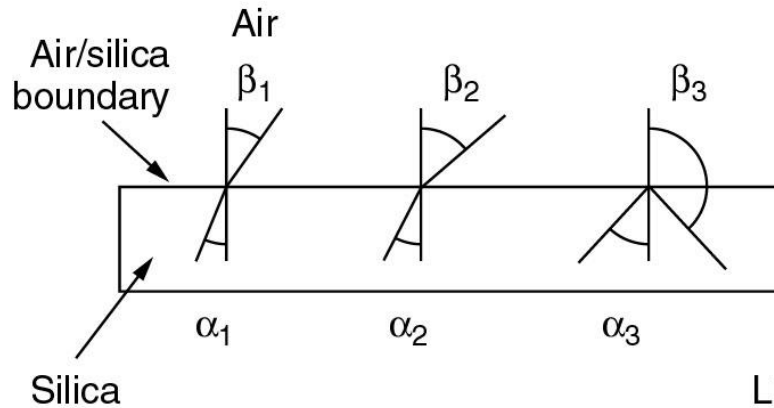
---

- ◆ High bandwidth, good immunity to noise
- ◆ High bandwidths (e.g. 1 GHz)
- ◆ Low attenuations

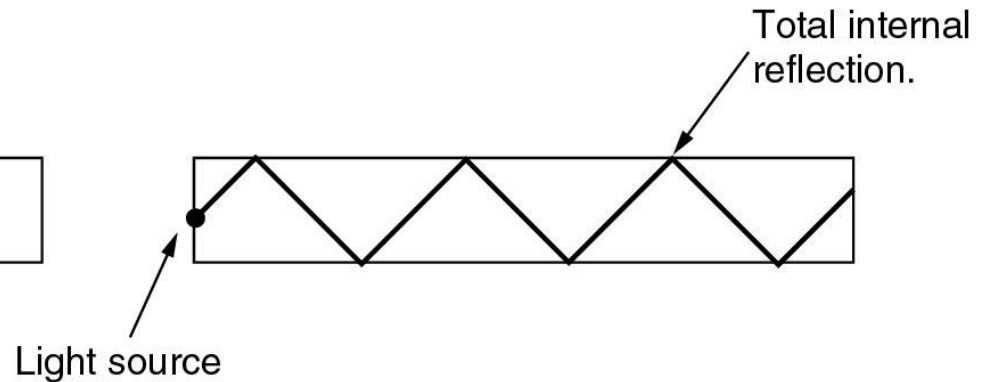


# Fiber Optics

---



(a)

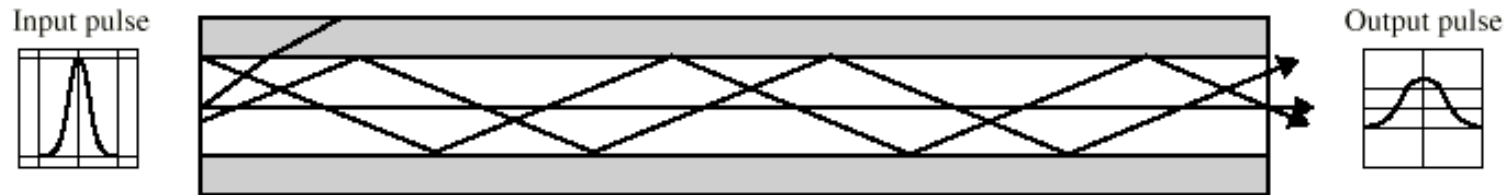


(b)

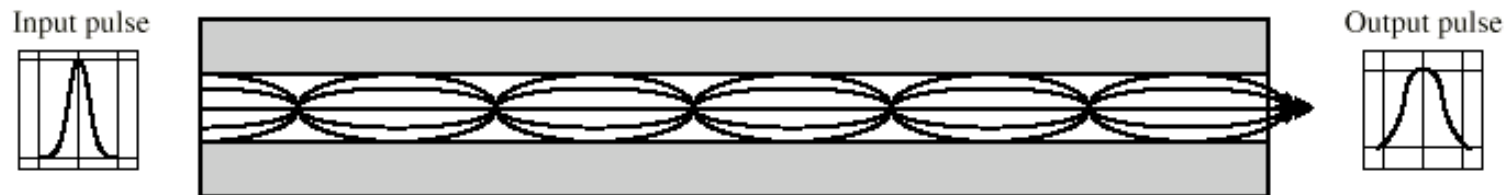
- (a) Three examples of a light ray from inside a silica fiber impinging on the air/silica boundary at different angles.
- (b) Light trapped by total internal reflection.

# *Fiber Optical – Multimode vs Monomode*

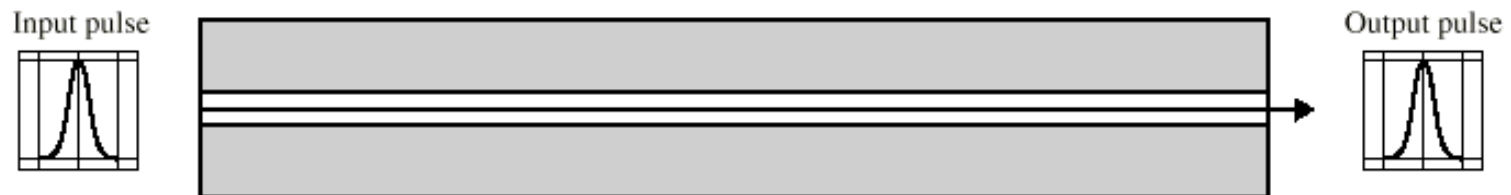
---



(a) Step-index multimode



(b) Graded-index multimode

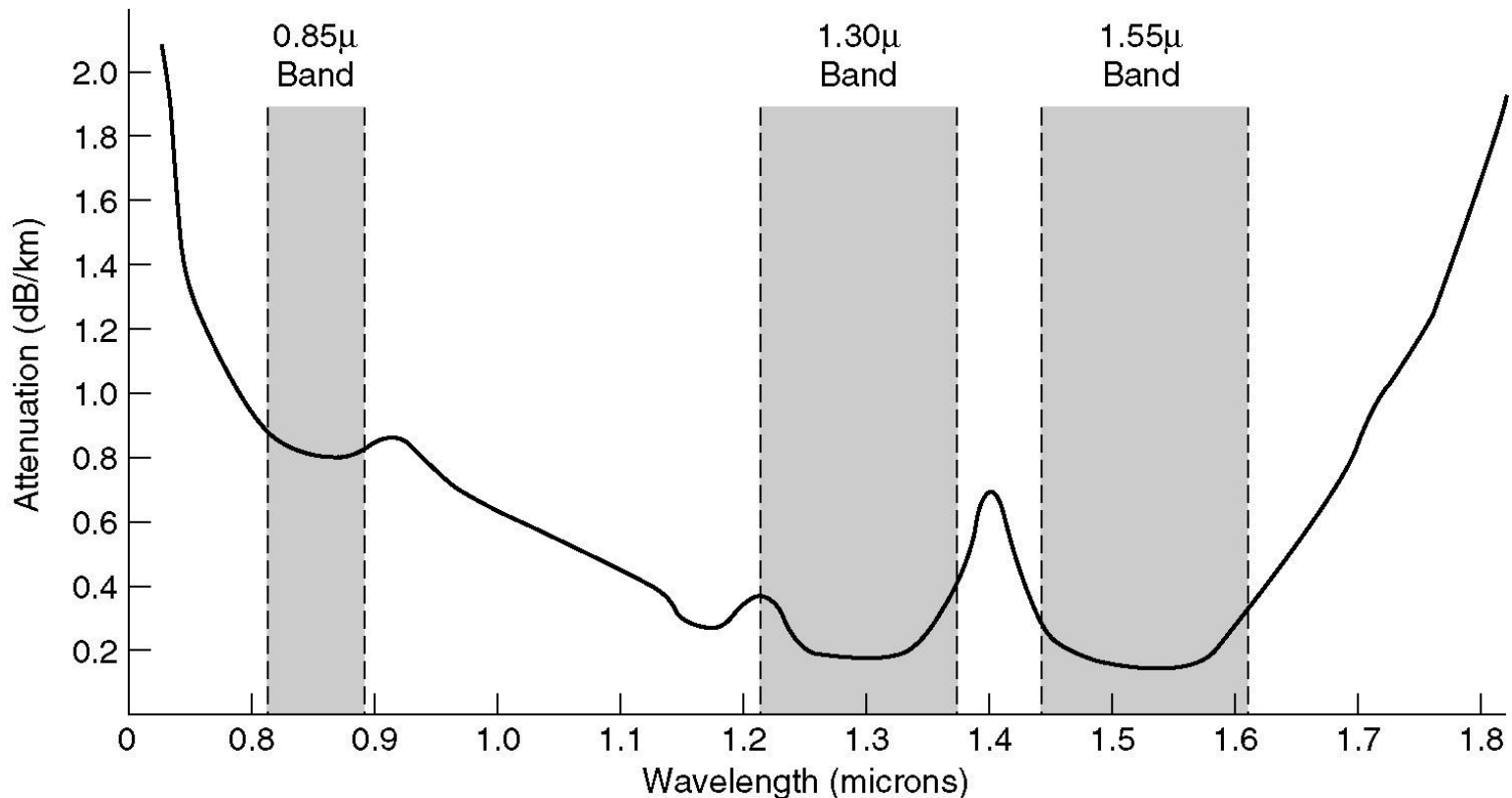


(c) Single mode

# *Optical Fiber*

---

- ◆ Attenuation of light through fiber in the infrared region
- ◆ Bandwidths of 30 000 GHz ! Very low attenuations  $< 1\text{dB/km}$
- ◆ Data transmission: Light (1) / No light (0)  $\rightarrow$  NRZ



# Wavelength( $\lambda$ ), Propagation Delay

$$\lambda = vT \quad \lambda f = v$$

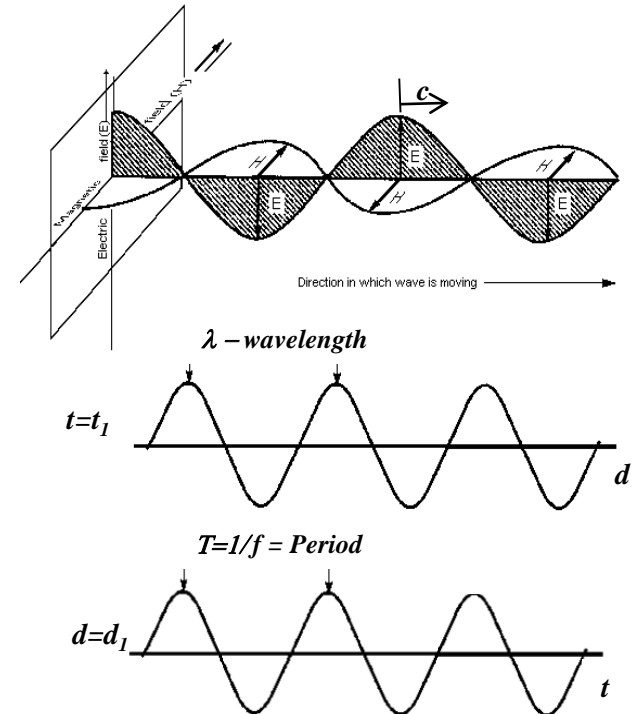
- $\lambda$ : wavelength
- $v$ : velocity of the wave
- $f$ : frequency

» Speed of light in free space  $c = 3 * 10^8$  m/s

» Propagation delays ( $\mu\text{s} / \text{km}$ )

- Free space ( $1/c$ ):  $3.3\mu\text{s} / \text{km}$
- Coaxial cable:  $4\mu\text{s} / \text{km}$
- UTP:  $5\mu\text{s} / \text{km}$
- Optical fiber:  $5\mu\text{s} / \text{km}$

speed decreases





# *Wireless Transmission*

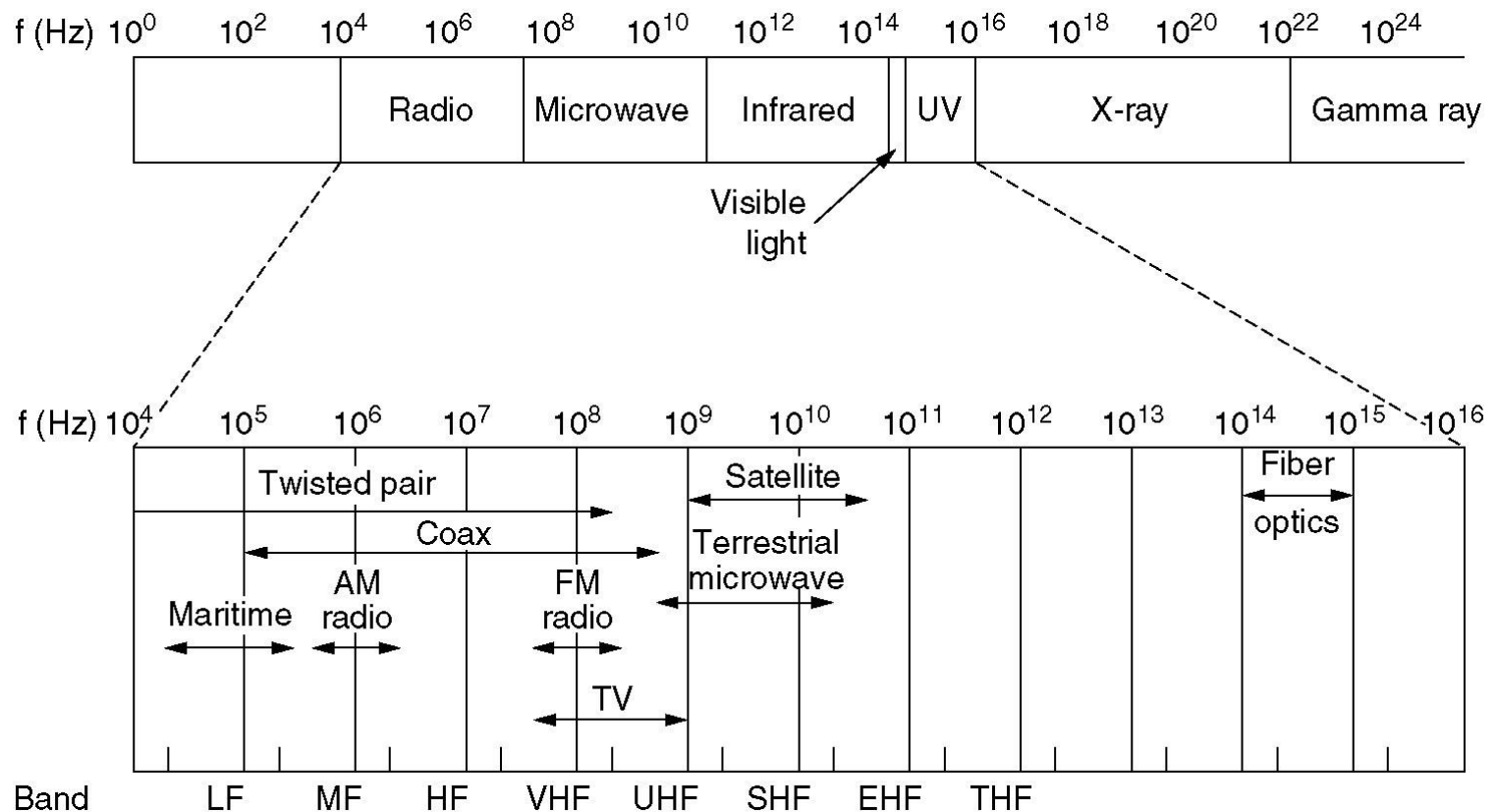
---

- ♦ The Electromagnetic Spectrum
- ♦ Radio Transmission

# *The Electromagnetic Spectrum*

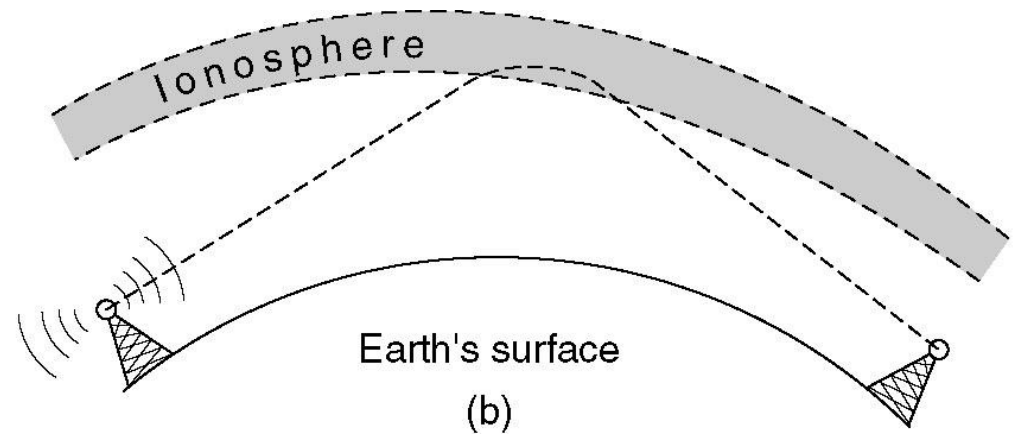
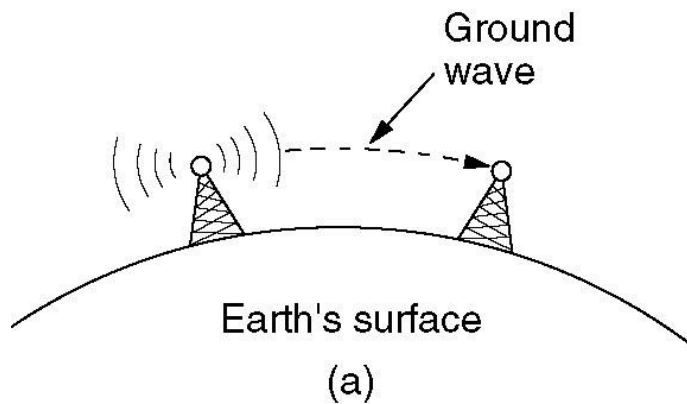
---

The electromagnetic spectrum and its uses for communication



# *Radio Transmission*

---



- (a) In the LF and MF bands, radio waves follow earth curvature
- (b) In the HF band, they bounce off the ionosphere.

## *To Think*

---

- ♦ How does the attenuation of an wireless channel vary with the distance?

## *Free Space Loss*

---

- ◆ Free space loss, ideal isotropic antenna

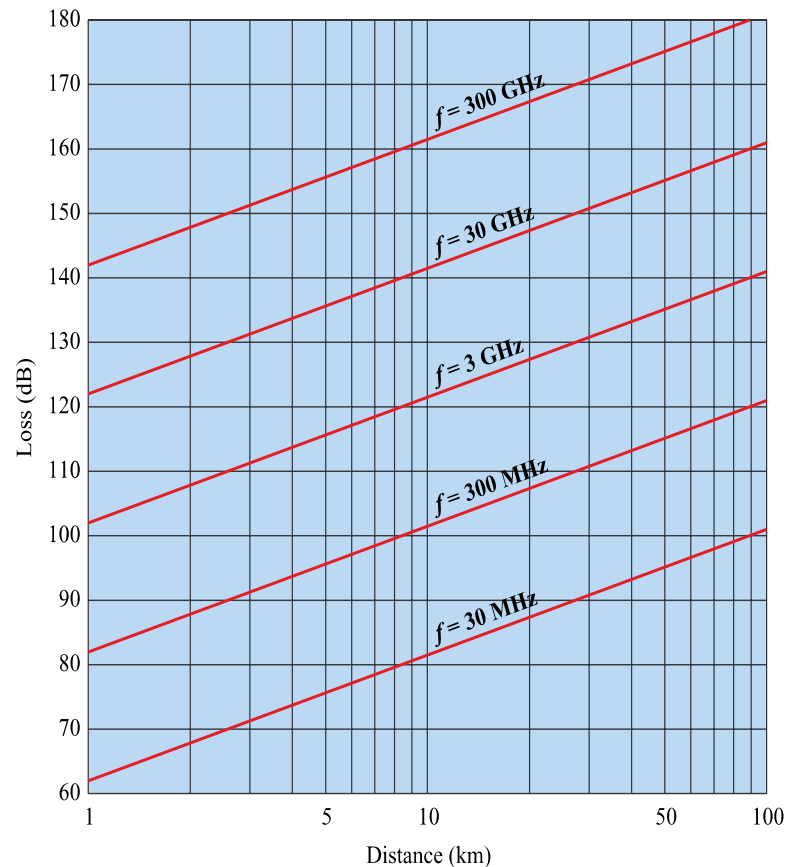
$$\frac{P_t}{P_r} = \frac{(4\pi d)^2}{\lambda^2} = \frac{(4\pi f d)^2}{c^2} \quad \lambda f = c$$

- $P_t$  = signal power at transmitting antenna
- $P_r$  = signal power at receiving antenna
- $\lambda$  = carrier wavelength
- $d$  = propagation distance between antennas
- $c$  = speed of light ( $3 \times 10^8$  m/s)

# *Free Space Loss, in dB*

---

$$L_{dB} = 10 \log \left( \frac{P_t}{P_r} \right) = 20 \log \left( \frac{4\pi f d}{c} \right) = 20 \log(f) + 20 \log(d) - 147.56 \text{ dB}$$



# *Homework*

---

## 1. Review slides

Important: slides do not address details (no time!). **Book(s) must be read!**

## 2. Read from Tanenbaum

» Sections 2.1, 2.2, 2.3, 2.5, 2.6, 2.8, 2.9

## 3. Read from Bertsekas&Gallager

» Sections 2.1, 2.2

## 4. Answer questions at moodle