



សាកលវិទ្យាល័យភូមិន្ទភ្នំពេញ  
Royal University of Phnom Penh

# Data Communication I

## Chapter 3.1 : Digital transmission

Lecturer: **CHHORN SYLUN**

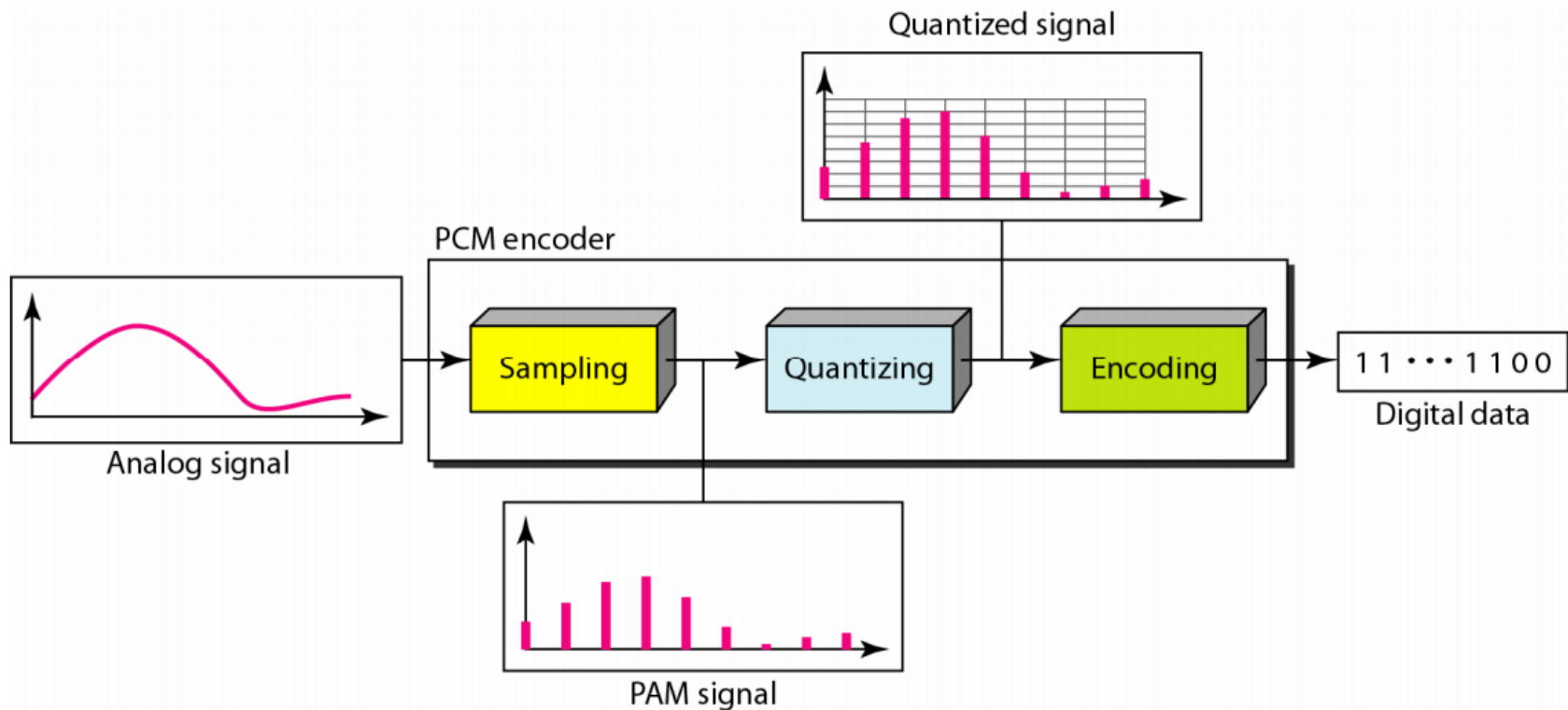
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# Outline

- ❑ Analog to digital conversion
  - Sampling
  - Quantization
  - Encoding
- ❑ Line Coding
- ❑ ADC Types
- ❑ Transmission Mode

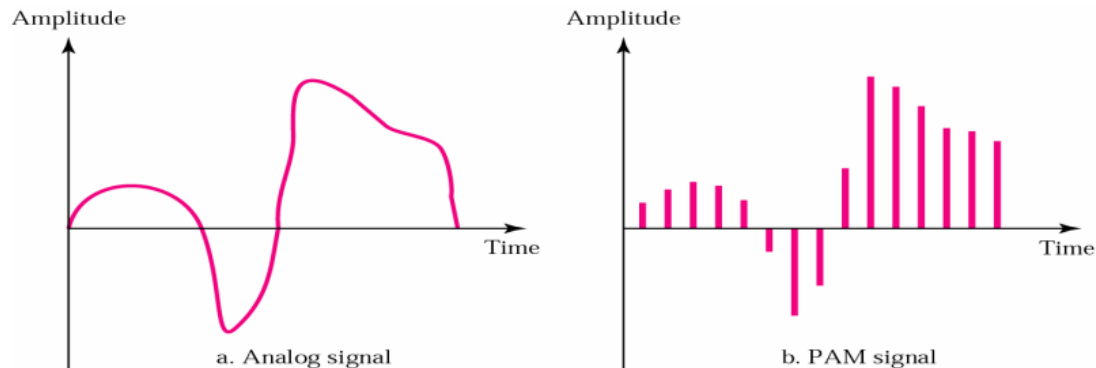
# Analog to digital conversion



*Summary scheme of Analog to Digital Conversion*

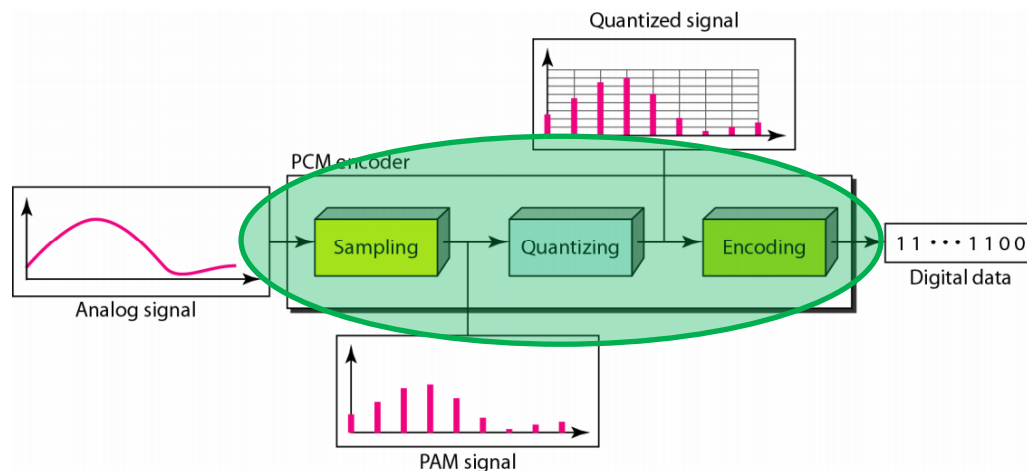
# Analog to digital conversion

- Microphone creates analog voice and Camera creates analog videos which is called analog data.
- To transmit this analog data over digital signals we need an **analog to digital conversion**.
- **Pulse Amplitude Modulation (PAM)**, takes an analog signal, samples it and generates a series of pulses based on the results of sampling.



# Analog to digital conversion

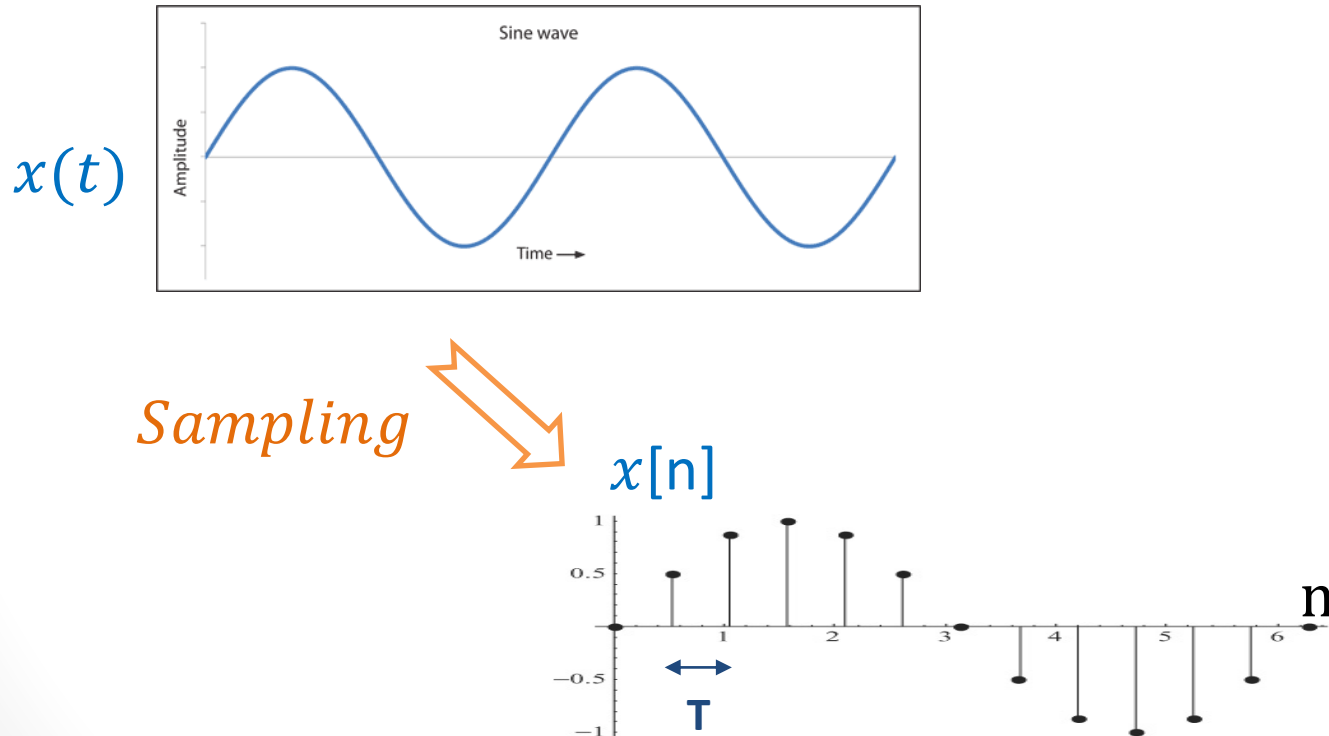
- However **PAM** is not useful in data communications because even though it translates the original waveform to a series of pulses, these pulses are still an analog.
- To make them digital we modify them using **Pulse Code Modulation (PCM)**.
- There are three steps in **PCM**: **Sampling**, **Quantization** and **Encoding**.



# Analog to digital conversion

## 1. Sampling

What is Sampling Mean?



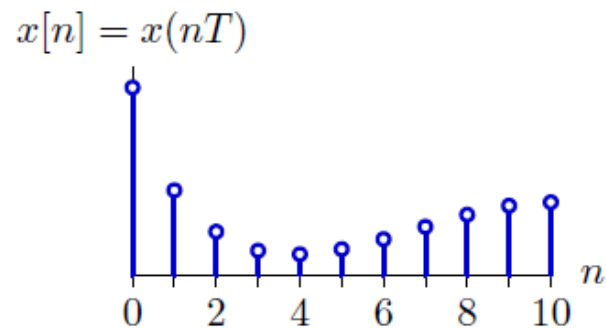
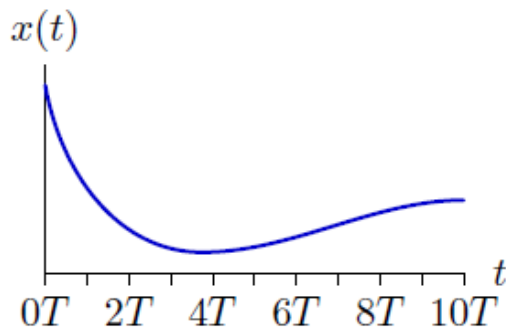
# Analog to digital conversion

1. **Sampling:** the process of convert analog signal into digital signal.

$x(t)$  is analog signals  
 $x[n]$  is digital signals

$$x[n] = x(t)|_{t=nT} = x(nT)$$

- $x(t)$ : analog signal
- $x[n]$ : digital signal
- $n$  : is sample number
- $T$  : is sampling interval



$T$  = sampling interval

# Analog to digital conversion

1. **Sampling:** the process of convert analog signal into digital signal.

**Example:**

1.  $x(t) = e^{at} \xrightarrow[t = nT]{} x[n] = e^{anT} = e^{\frac{an}{F_s}}$

*time* (pointing to  $x(t)$ )

Sample number [0, 1, 2, 3,.....] (pointing to  $n$ )

Sampling Period (T) (pointing to  $T$ )

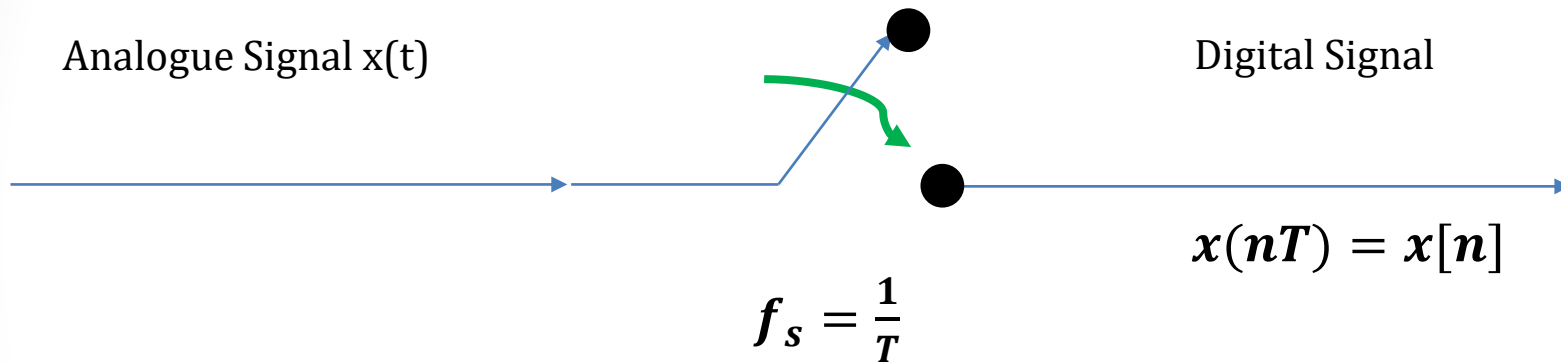
2.  $x(t) = 10e^{-t} - 5e^{-0.5t} \xrightarrow[t = nT]{} x[n] = 10e^{-nT} - 5e^{-0.5nT}$

$= 10e^{\frac{-n}{F_s}} - 5e^{\frac{-0.5n}{F_s}}$



# Analog to digital conversion

1. **Sampling:** the process of convert analog signal into digital signal.



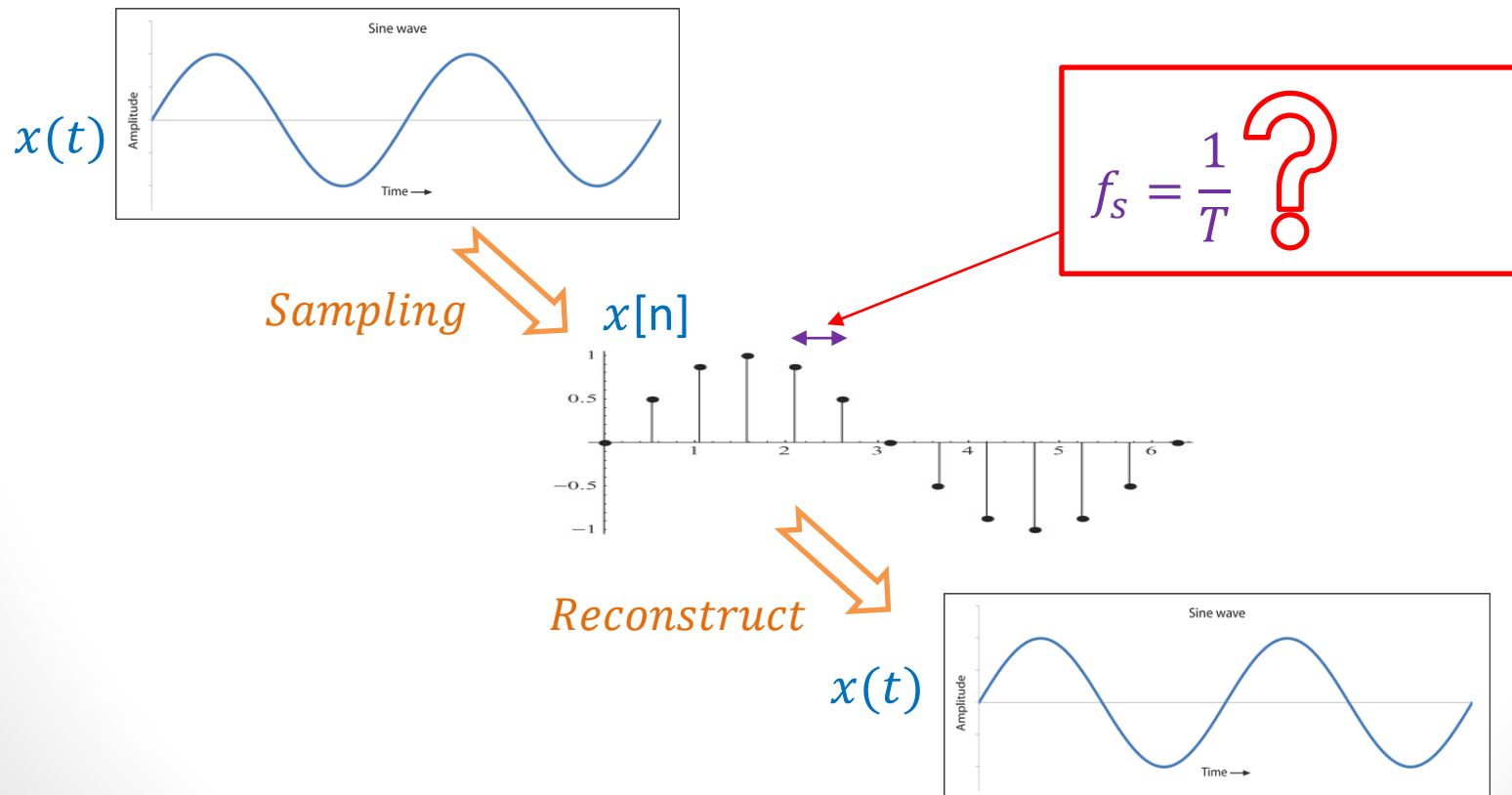
*The constant  $T$  is the sampling interval or period and the sampling frequency:*

$$f_s = \frac{1}{T} \text{ Hz}$$

# Analog to digital conversion

1. **Sampling:** the process of convert analog signal into digital signal.

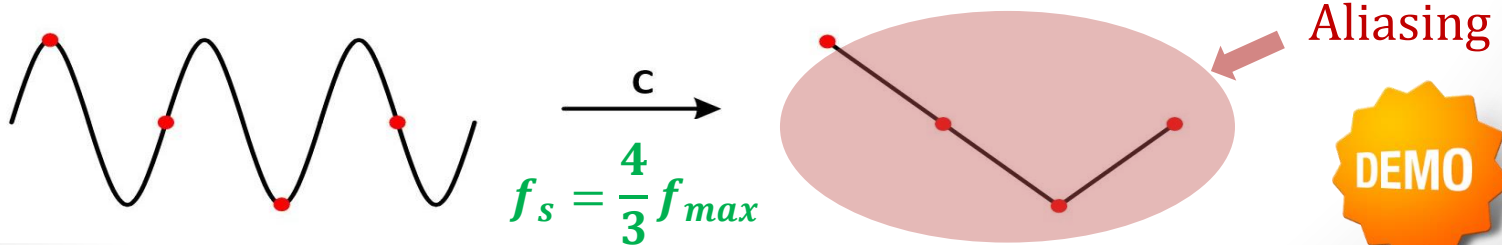
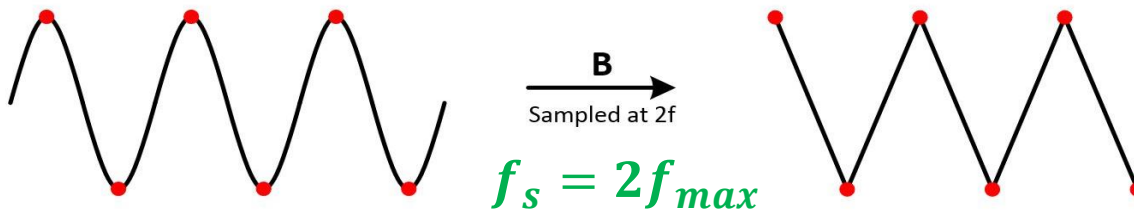
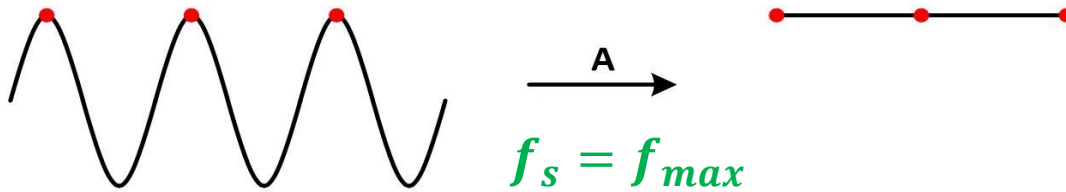
How to choose  $F_s$ ?



# Analog to digital conversion

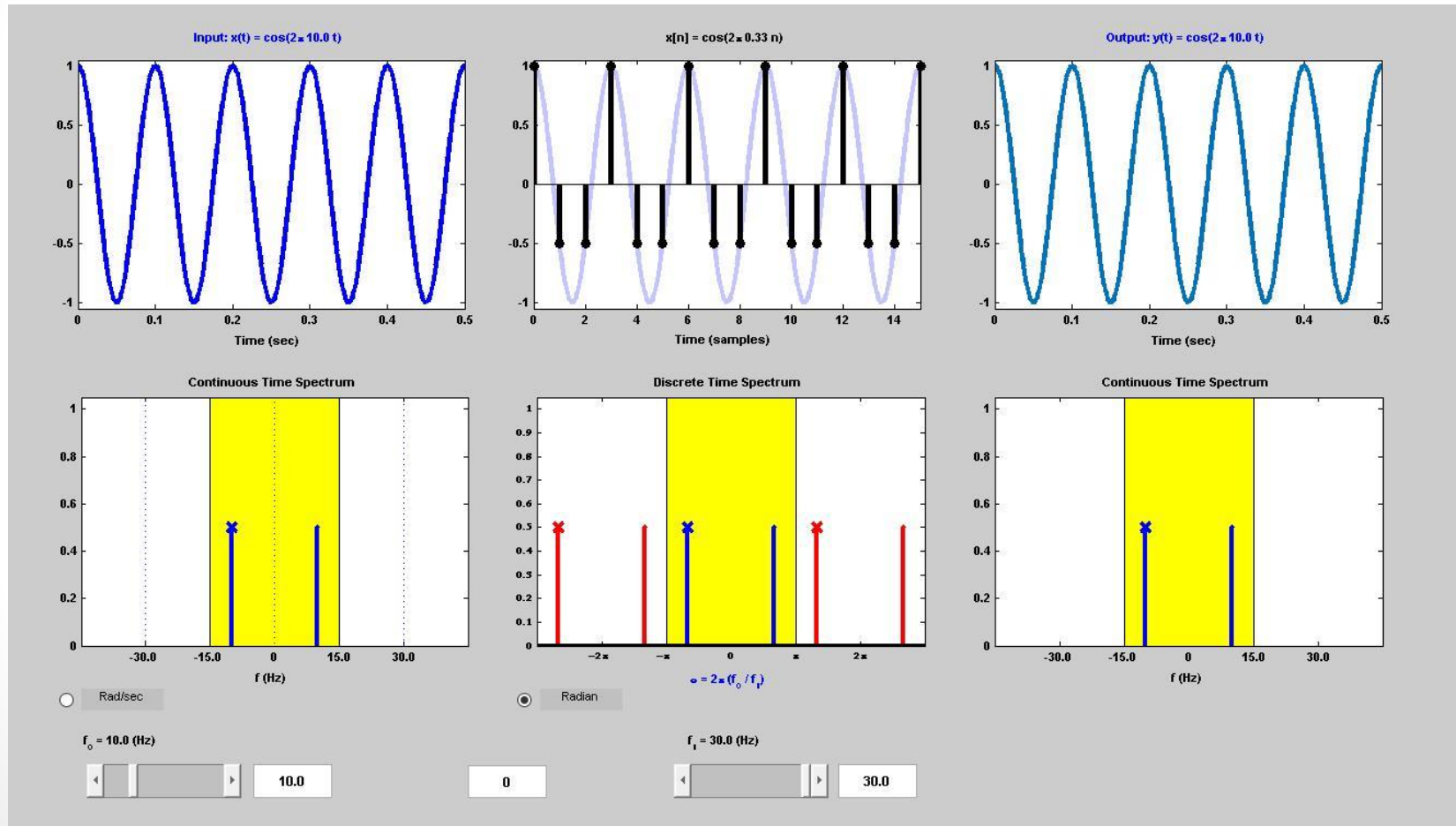
1. **Sampling:** the process of convert analog signal into digital signal.

How to choose  $F_s$ ?



# Analog to digital conversion

## Sampling Application Demo



# Analog to digital conversion

1. **Sampling:** the process of convert analog signal into digital signal.

## Nyquist Theorem

- If the highest frequency contained in an analog signal  $x(t)$  is  $f_{max}$  then
- The signal must be sampled at the following rate
$$f_s \geq 2f_{max}$$
- Hence Analog Signal  $x(t)$  can be exactly recovered from its sample values using an interpolation function.

# Analog to digital conversion

1. **Sampling:** the process of convert analog signal into digital signal.

**Example:** Audio CDs use a sampling rate,  $f_s$  of  $44.1kHz$  for storage of the digital audio signal.

- This sampling frequency is slight more than  $2f_{max}$  [ $f_{max} = 20kHz$ ], which is generally accepted upper limit of human hearing and perception of music sound.
- Human hearing:  $20Hz < f < 20kHz$



# Analog to digital conversion

1. **Sampling:** the process of convert analog signal into digital signal.

## Example

Define the maximum frequency for below signal:

a)  $x(t) = 2\sin(2\pi 200t) + \cos(300\pi t)$

b)  $x(t) = 5\cos(200\pi t) + \cos(300t) + \cos\left(\frac{1000}{\pi}t\right)$

# Analog to digital conversion

1. **Sampling:** the process of convert analog signal into digital signal.

## Exercise

Consider the analog signal:  $x(t) = 3\cos 100\pi t$

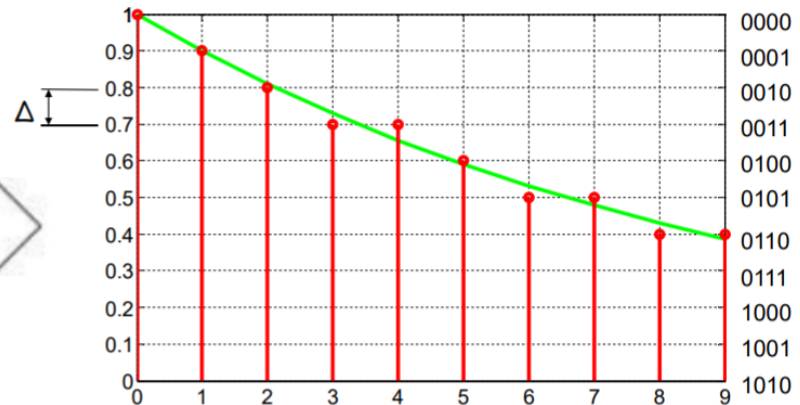
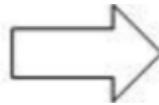
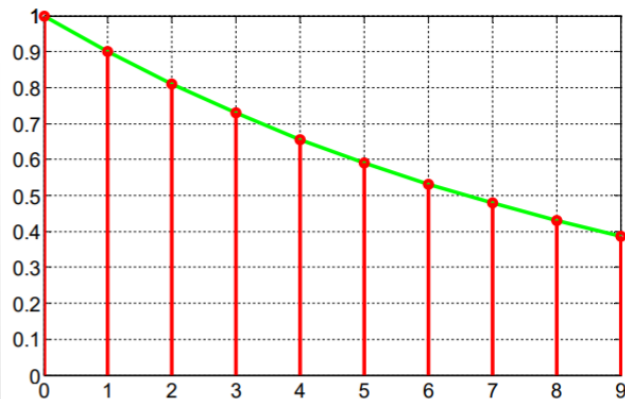
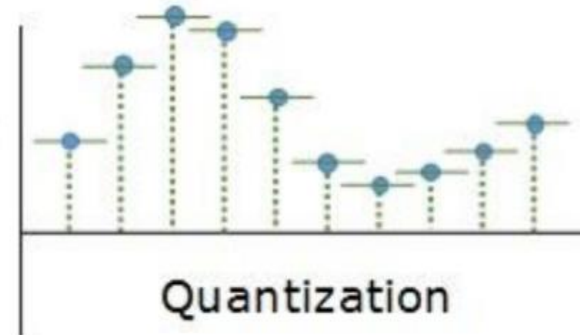
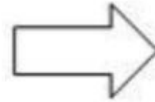
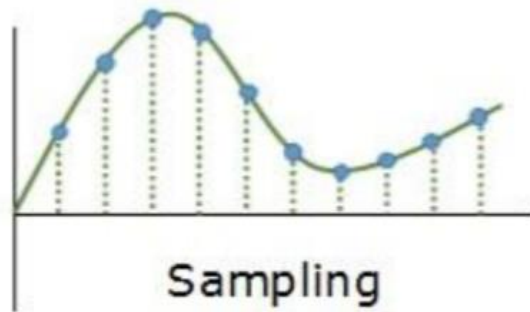
- a) Determine sampling rate required to avoid aliasing.
- b) Suppose that the signal is sampled at the rate of  **$F_s = 200 \text{ Hz}$** , what is the discrete time signal obtained after sampling?
- c) Suppose that the signal is sampled at the rate of  **$F_s = 75 \text{ Hz}$** , what is the discrete time signal obtained after sampling?



# Analog to digital conversion

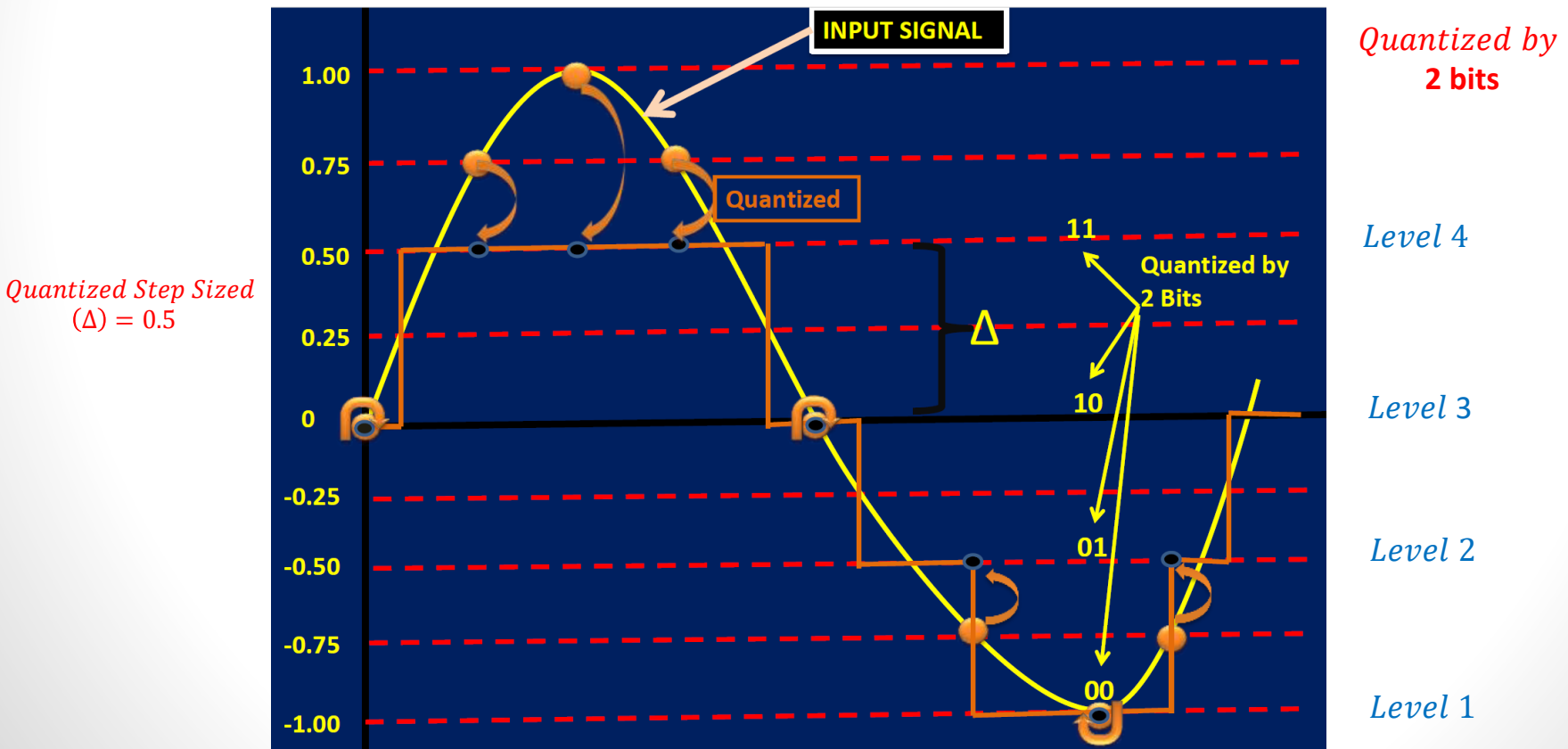
## 2. Quantization

What is Quantization mean?



# Analog to digital conversion

2. **Quantization:** the process whereby a continuous of amplitude values is represented by a finite set of discrete value. Quantization will round off the samples to the nearest quantization level



# Analog to digital conversion

2. **Quantization:** the process whereby a continuous of amplitude values is represented by a finite set of discrete value. Quantization will round off the samples to the nearest quantization level.

- **n=2 bits** mean **L=4 Levels** in quantization
- Quantized step sided ( $\Delta$ ) or Resolution of quantizer( $q$ )

$$(\Delta) = (q) = \frac{\{X_{(max)} - X_{(min)}\}}{L} = \frac{\{X_{(max)} - X_{(min)}\}}{2^n}$$

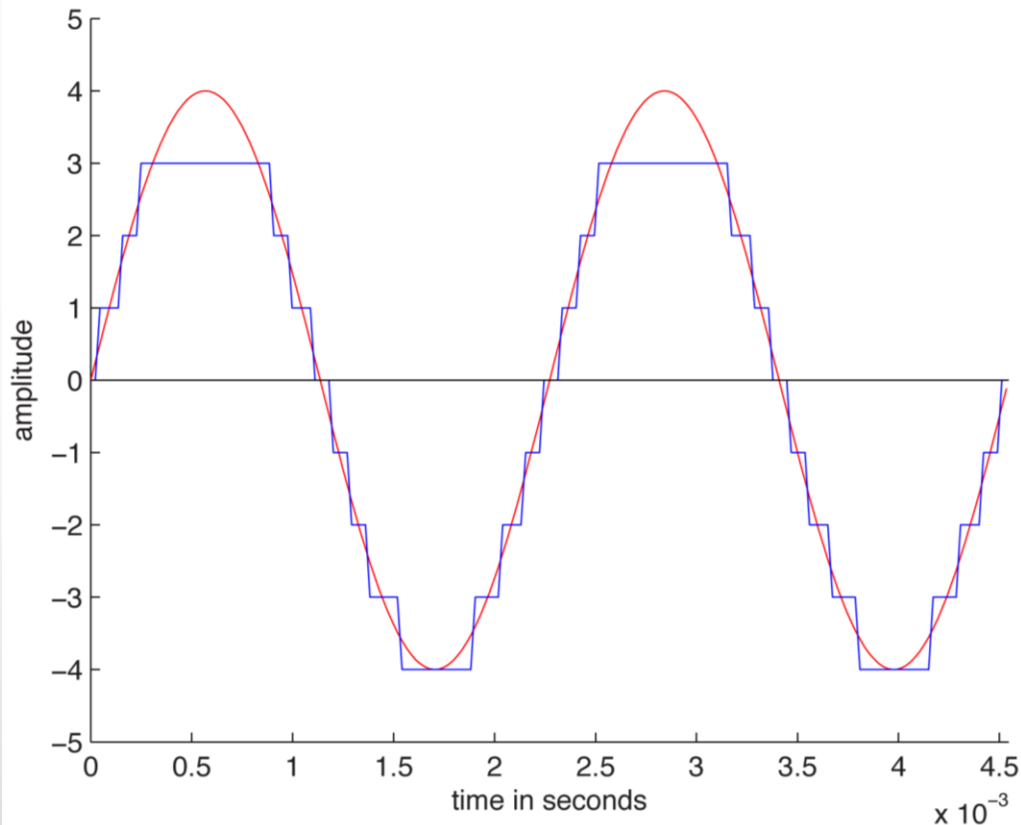
- From previous slide:

$$\text{Quantized step sided } (\Delta) = \frac{\{1 - (-1)\}}{4} = 0.5V$$

# Analog to digital conversion

**Example:** Answer the questions from looking the graph

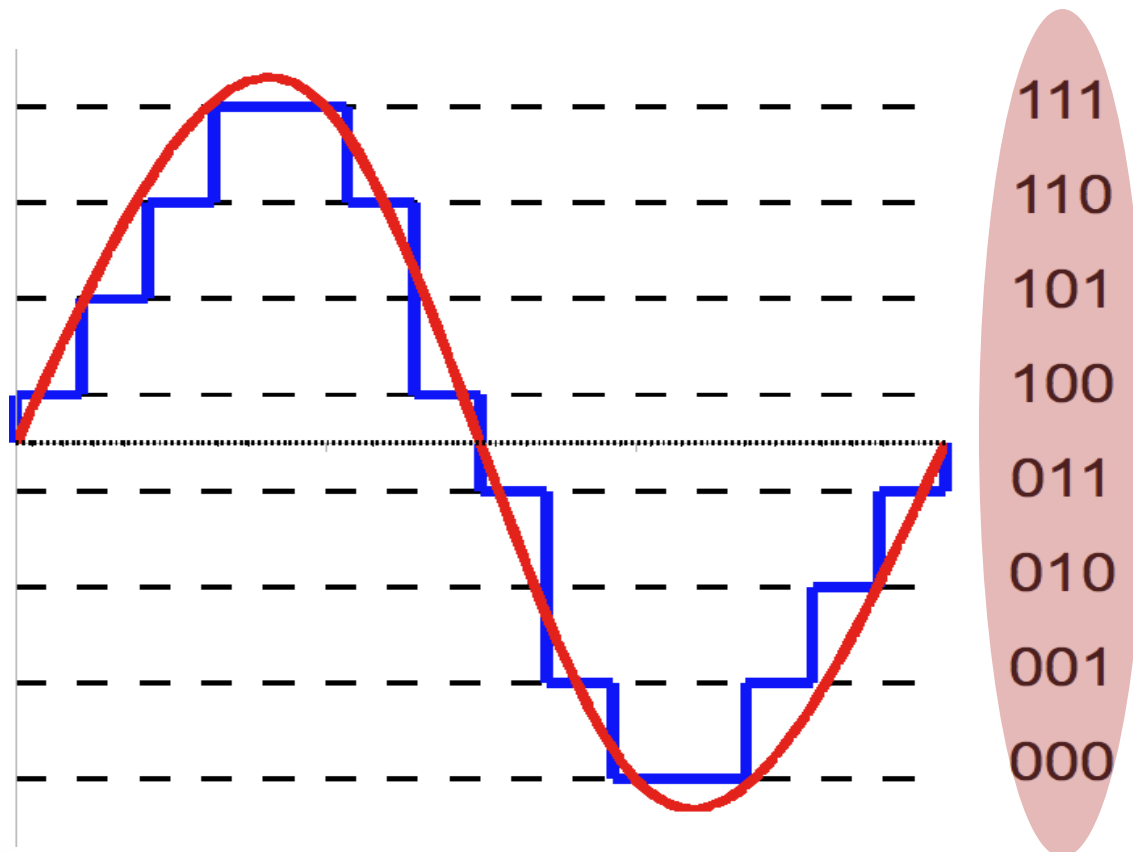
- How many level in the quantizing?
- Find Quantized step sided ( $\Delta$ )?



# Analog to digital conversion

## 3. Encoding:

What is Encoding mean?



# Analog to digital conversion

3. **Encoding:** (Binary Encoding) will convert each quantized value into a codeword consisting of binary bits.

- Binary encoding for 4 levels:

**00, 01, 10, 11**

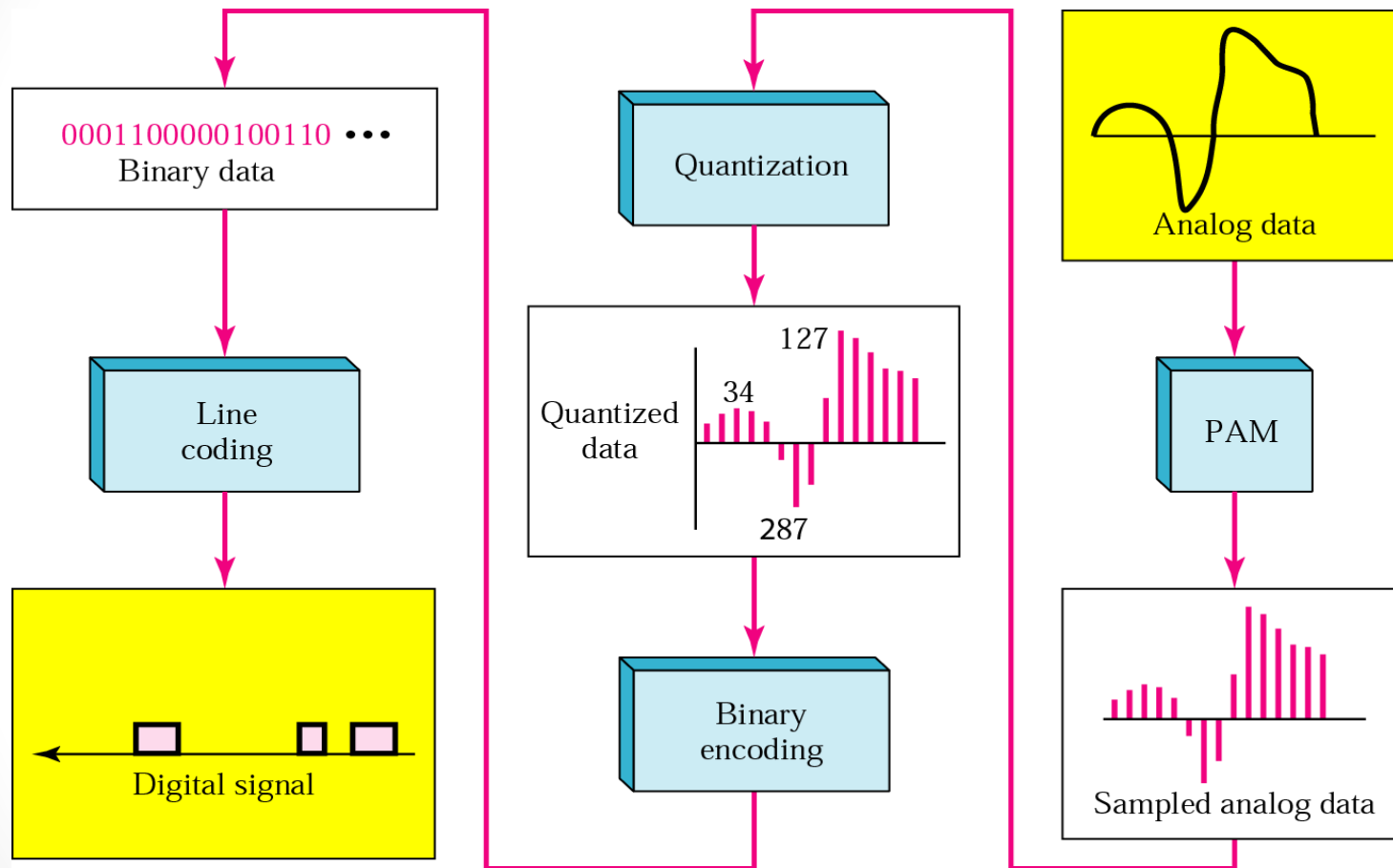
- Binary encoding for 8 levels:

**000, 001, 010, 011, 100, 101, 110, 111**

- **Example**

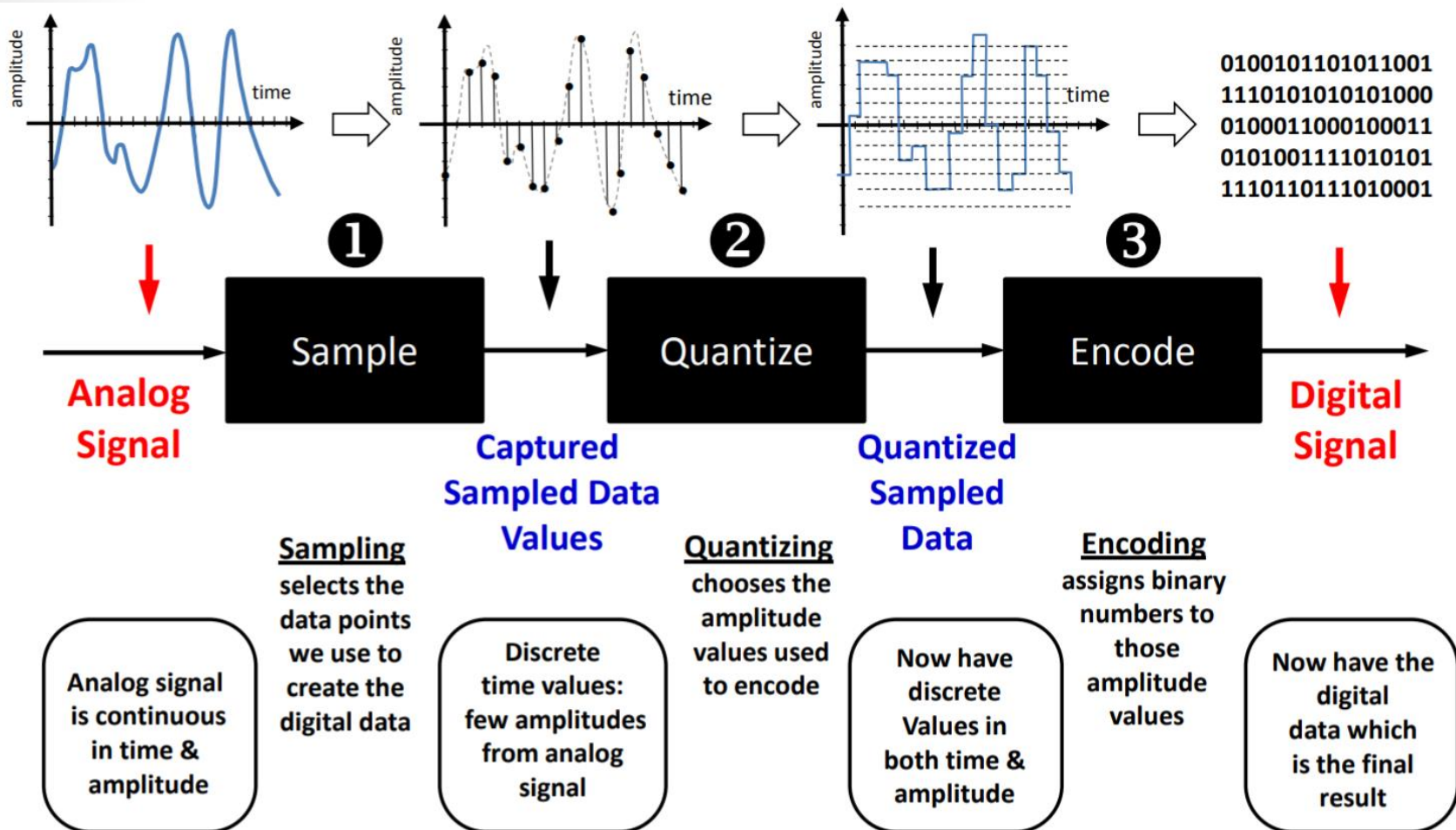
Original sampling signal	0	0.707	1	0.707	0	-0.707	-1	-0.707
Quantized value	0	0.5	0.5	0.5	0	-0.5	-1	-0.5
Binary value	10	11	11	11	10	01	00	01

# Analog to digital conversion



*Summary scheme of Analog to Digital Conversion 1*

# Analog to digital conversion



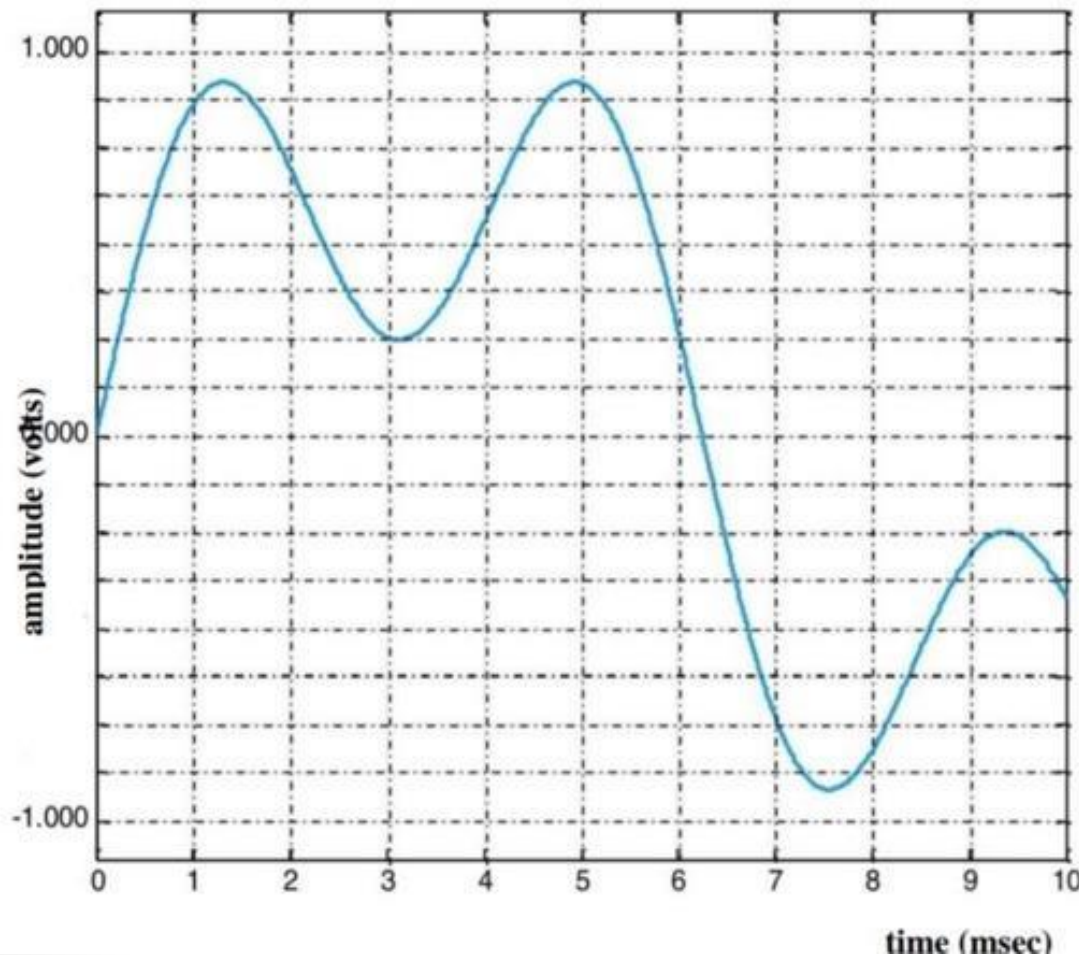
*Summary scheme of Analog to Digital Conversion 2*



# ADC: Exercise

Consider the following analog waveform. This waveform is sampled at a 500 Hz rate and quantized with a 2-bit quantizer (i.e., A/D converter). The input range is -1.0 to +1.0 V.

- Circle the sample points (first sample is at time  $t = 0$  sec).
- Indicate the quantization intervals and corresponding digital words.
- Indicate the digital word assigned to each sample point.
- What is the stream of binary bits generated after the A/D conversion is complete?
- What is the resulting bit rate from this A/D?

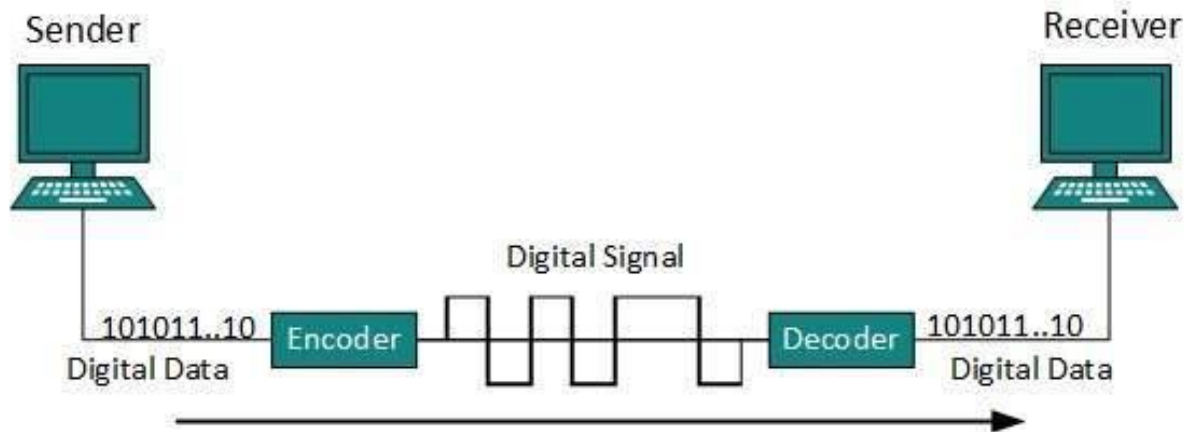


# Where we are now?

- ❑ Analog to digital conversion
- ❑ Line Coding
- ❑ Transmission Mode

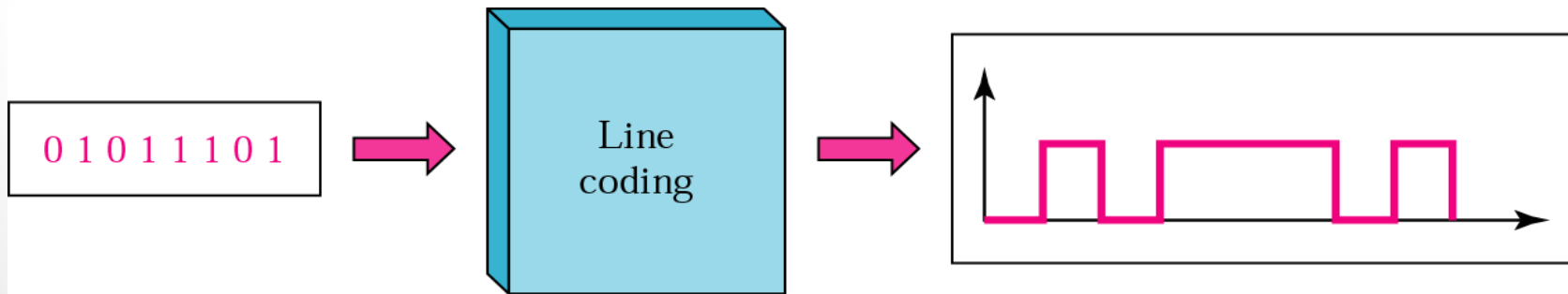
# Line Coding

- Data or information can be stored in two ways, analog or digital. Like data, signals can also be in analog or digital form.
- **Digital data** must convert into **digital signal** before transmit to another channel.
- To do this, we will use **Line Coding**.



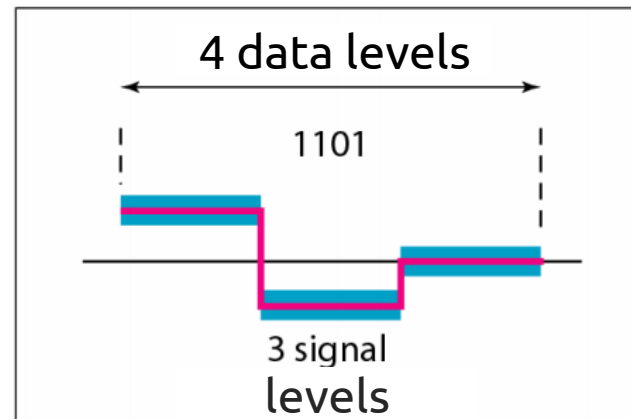
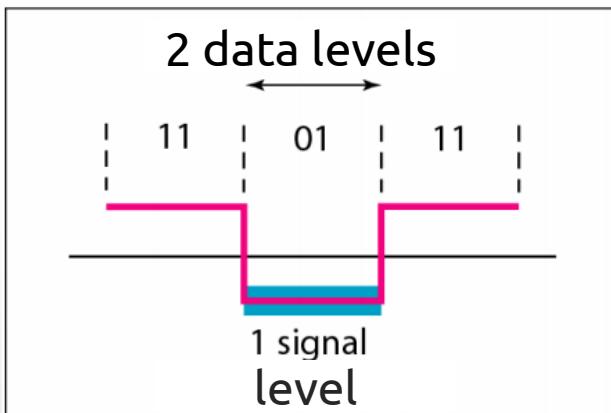
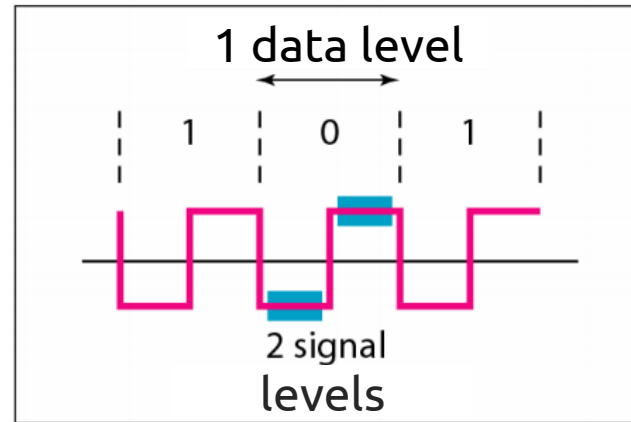
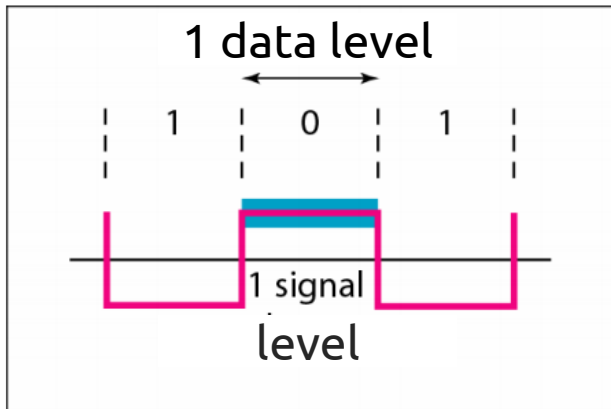
# Line Coding

- A line coding is the code used for data transmission of a digital signal over a transmission line.
- The process of coding is chosen so as to avoid overlap and distortion of signal such as inter-symbol interference.



# Line Coding

- **Signal Level:** number of values allowed in a particular signal
- **Data Level:** number of values used to represent data (**1 or 0**)



# Line Coding

- **Pulse Rate:** define the number of pulse per second  
A pulse is the minimum amount of time required to transmit a symbol. (baud or pulse per second)
- **Bit Rate:** define the maximum number of bit per second. (bit per second)
- The relationship between Pulse rate and Bit rate:

$$\text{Bit Rate} = \text{Pulse Rate} \times \log_2 L$$

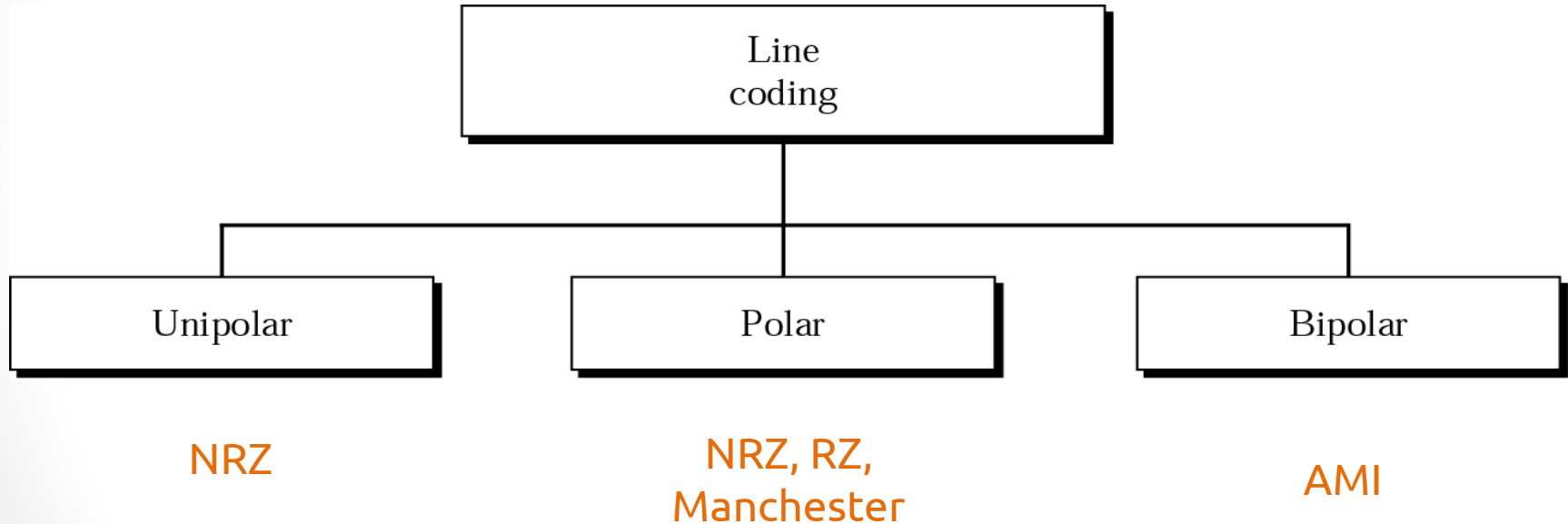
- **L** is the number of data levels of the signal
- If a pulse carries only 1 bit  $\rightarrow$  pulse rate = a bit rate
- If a pulse rate carries  $>1$  bit  $\rightarrow$  a bit rate is greater then the pulse rate

# Line Coding

**Example:** A signal has four data levels with a pulse duration of 1ms. Find the Pulse Rate and Bit Rate?

# Line Coding

- There are **three** types of line coding:

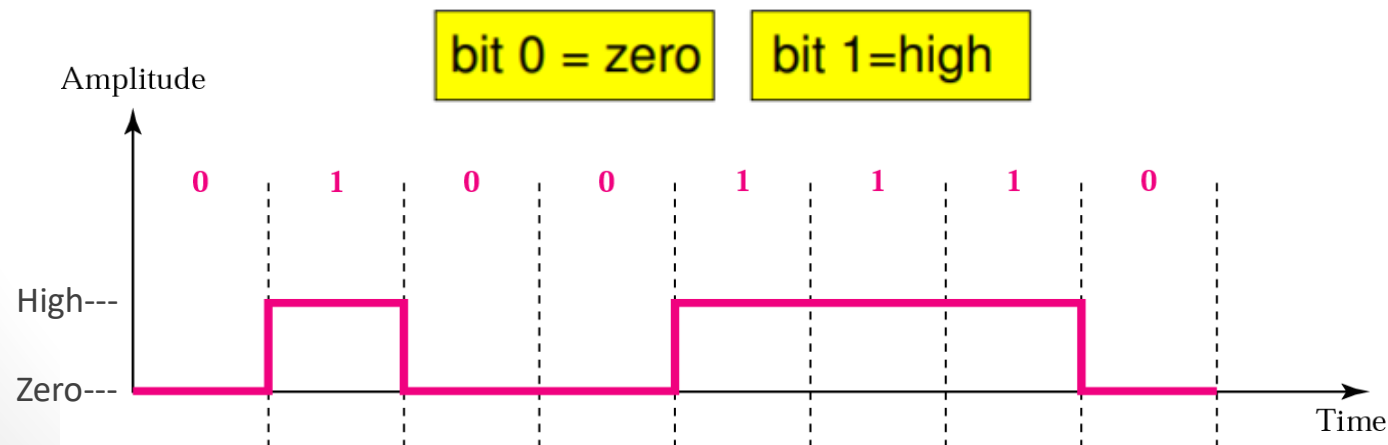




# Line Coding

## 1. Unipolar

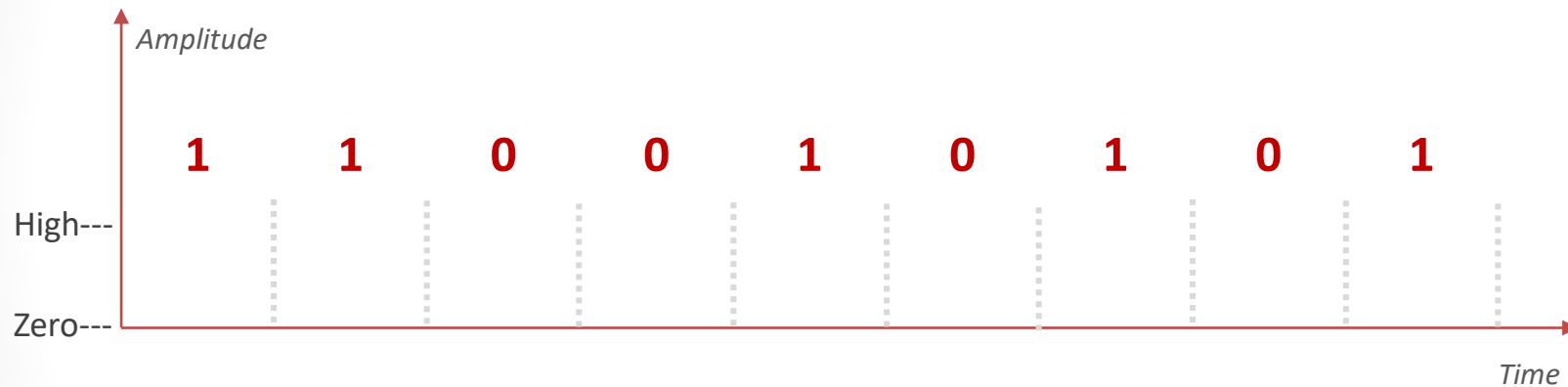
- Unipolar encoding schemes uses single voltage level to represent data.
- In this case, to represent binary **1** High voltage is transmitted and to represent **0** no voltage is transmitted.
- It is also called **Unipolar NRZ** ( Non Return Zero)



# Line Coding

## 1. Unipolar

**Example:** Draw the Unipolar of below bits stream



# Line Coding

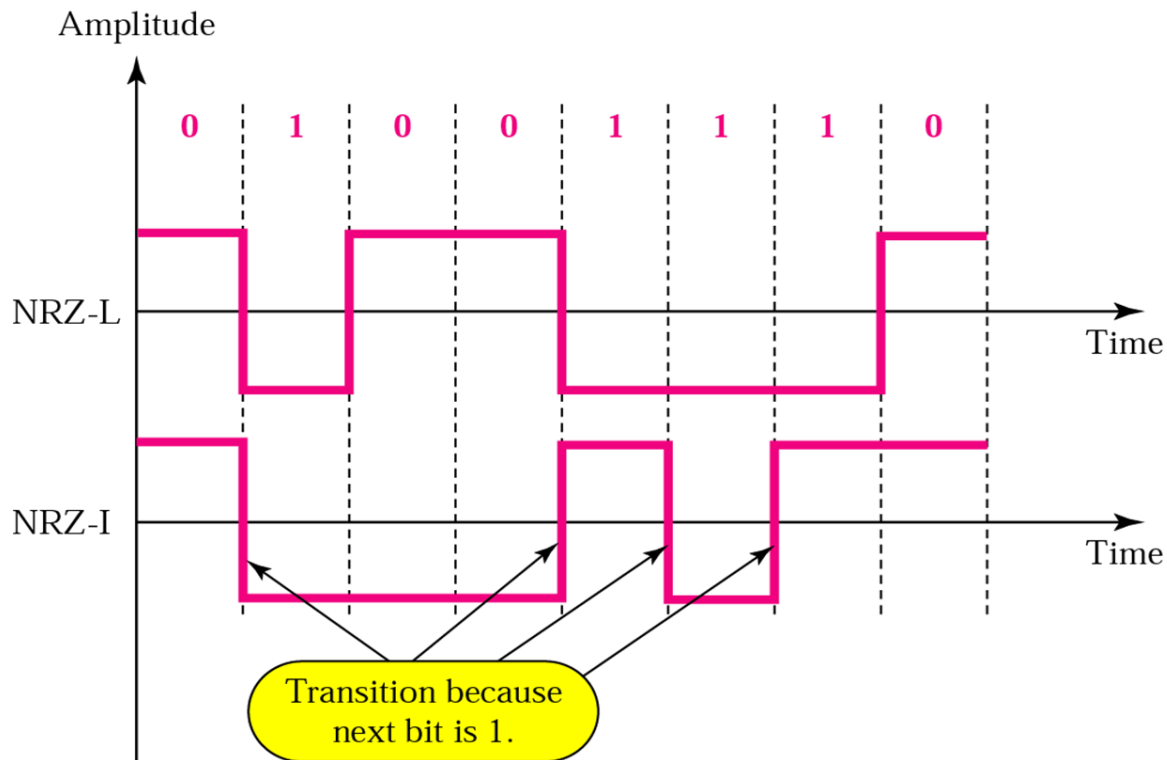
## 2. Polar

- Polar encoding schemes multiple voltage levels are used to represent binary values.
- Polar encodings are available in four types:
  1. NRZ-L
  2. NRZ-I
  3. RZ
  4. MANCHESTER

# Line Coding

## 2. Polar

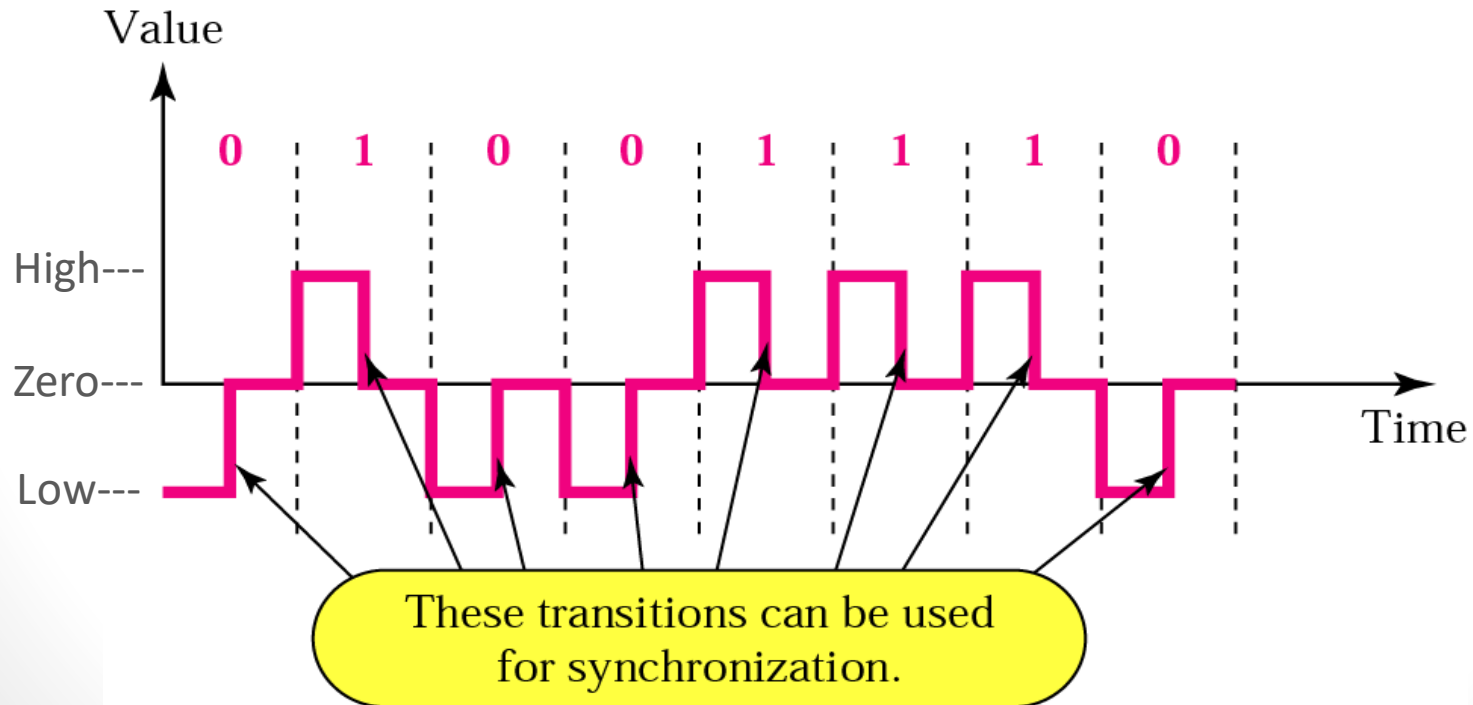
- **NRZ-L** encoding normally *bit 0= High, bit 1= Low*
- **NRZ-I** encoding the signal is *inverted if 1 is encountered*



# Line Coding

## 2. Polar

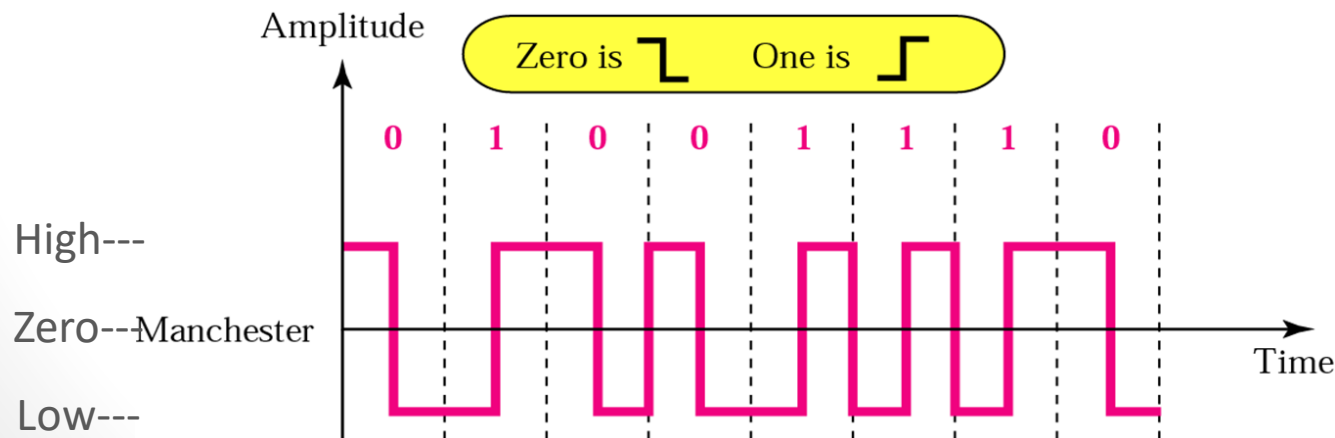
- **RZ** encoding the signal changes during each bit, not between bits: *bit 1 = high-to-zero*  
*bit 0 = low-to-zero*



# Line Coding

## 2. Polar

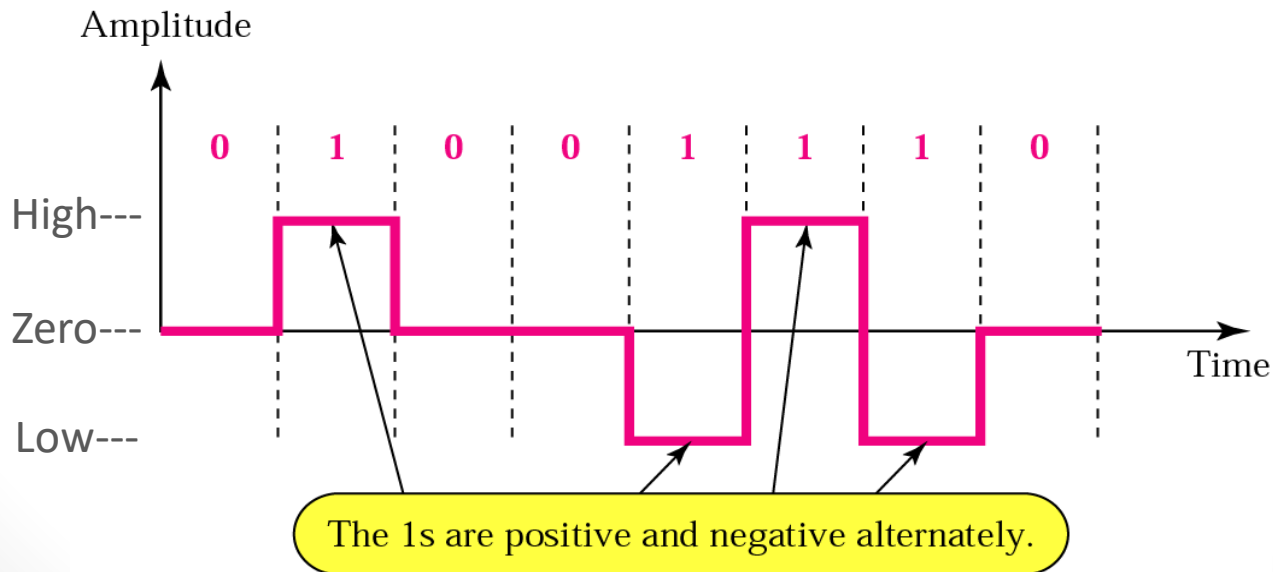
- **MANCHESTER**, this encoding scheme is a combination of RZ and NRZ-L.
- The transitions is at the middle of the bit and changes phase when difference bit is encountered.
- MANCHESTER encoding: *bit 1= low-to-high*  
*bit 0= high to low*



# Line Coding

## 3. Bipolar

- Bipolar encoding uses three voltage levels: positive, negative and zero.
- Zero voltage represents binary 0 and bits 1 is represented by alternating positive and negative voltages which is called **AMI** (Alternative Mark Inversion).



# Line Coding

## Exercise

NRZ-I and AMI start with High

	1	1	0	1	1	0	0	0	0	0	0	0	0	1
Unipolar NRZ														
NRZ-L														
NRZ-I														
RZ														
Manchester														
AMI														



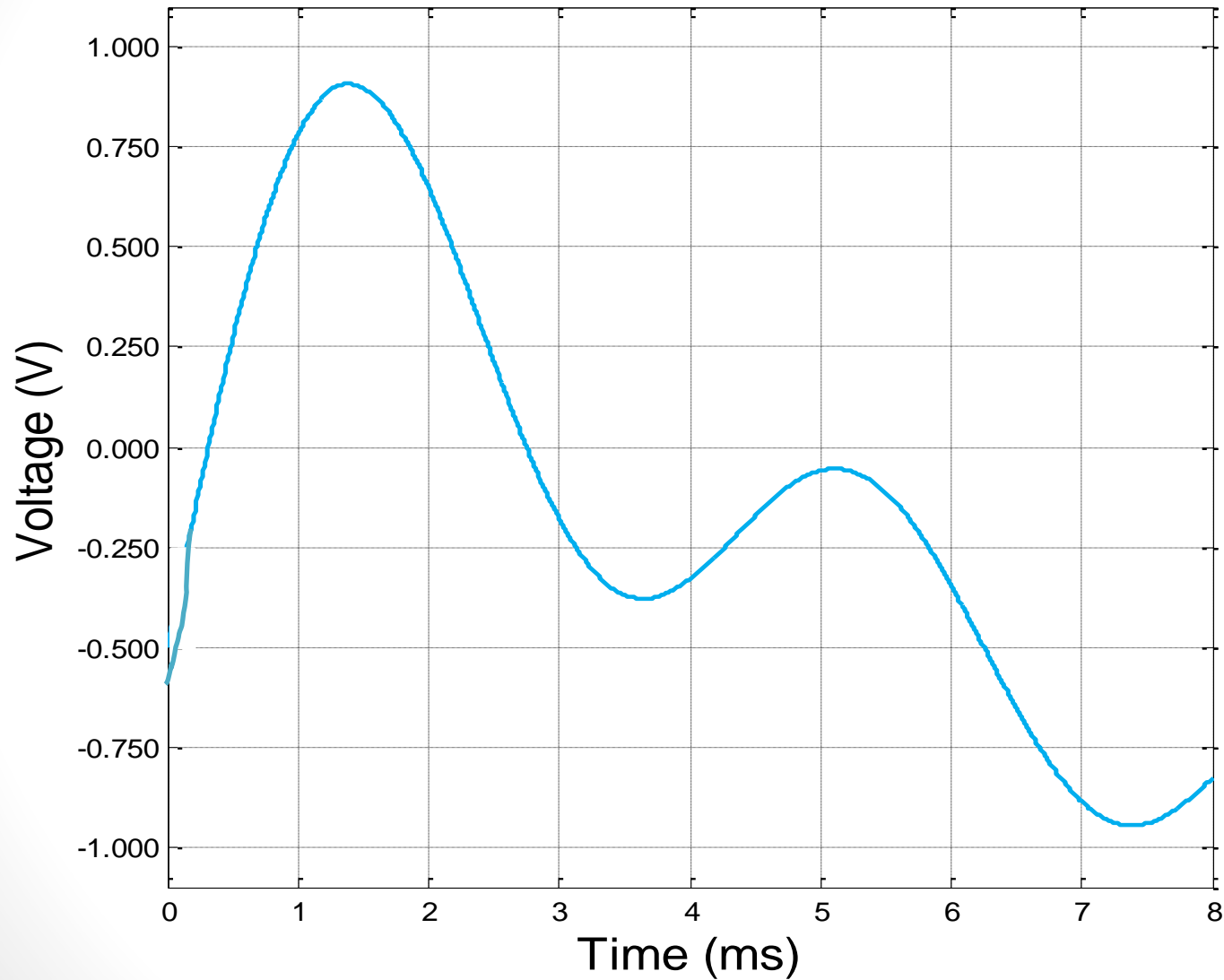
# Line Coding

## Exercise

Consider the following analog wave form. This waveform is sampled at a 1KHz rate and Quantized with a 3 bit quantizer (i.e., A/D converter). The input range is -1.0 to +1.0 V. Answer the following question:

- a) What is the resolution of this quantizer?
- b) Circle the sample points (first sample is at time  $t = 0s$ )
- c) Indicate the quantization intervals and corresponding digital words.
- d) Indicate the digital word assigned to each sample point.
- e) What is the stream of binary bits generated after A/D conversion is complete?
- f) What is the resulting bit rate from this A/D conversion?

# Line Coding



# Line Coding

## Exercise

Resulting bit stream from previous exercise find the digital signal using the following techniques: **NRZ-I and AMI start with High**

Bit stream are:

## Unipolar

## NRZ

## NRZ-L

## NRZ-I

RZ

# Manchester

**AMI**

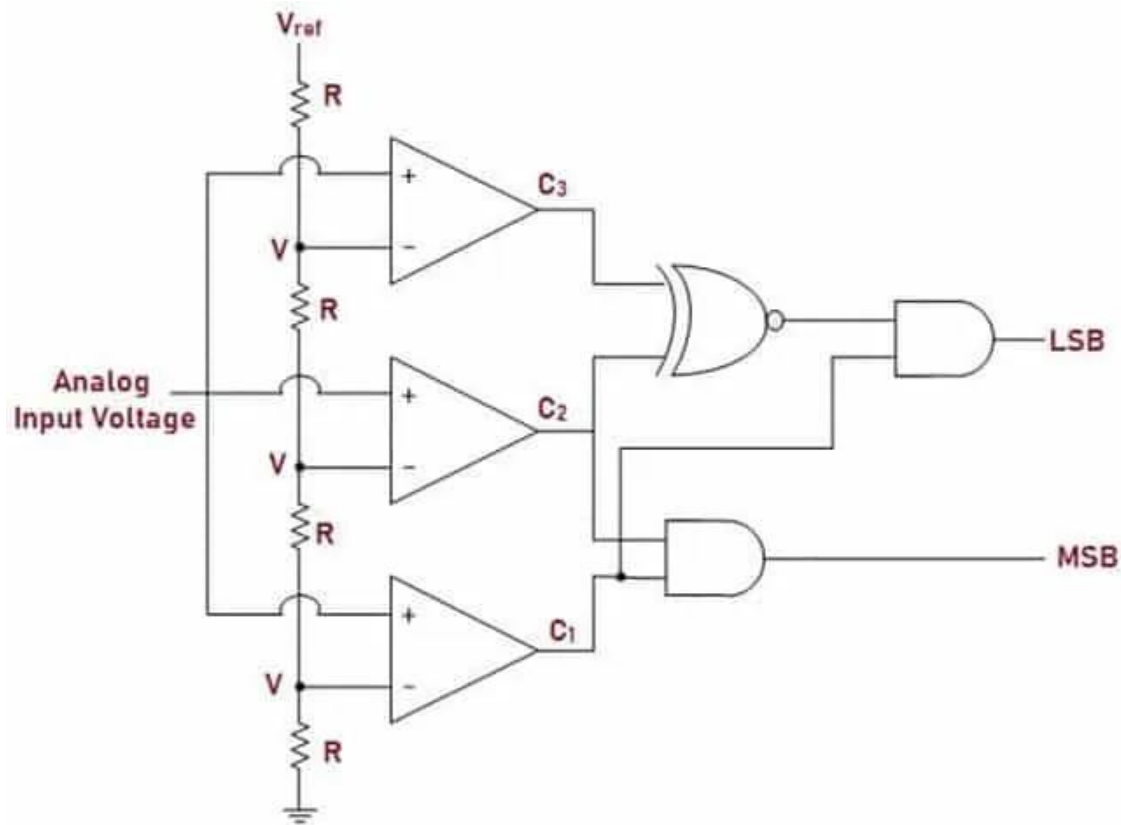
# ADC Types

The difference type of ADC are:

1. Flash ADC
2. Pipelined ADC
3. Sigma Delta ADC
4. Successive Approximation ADC

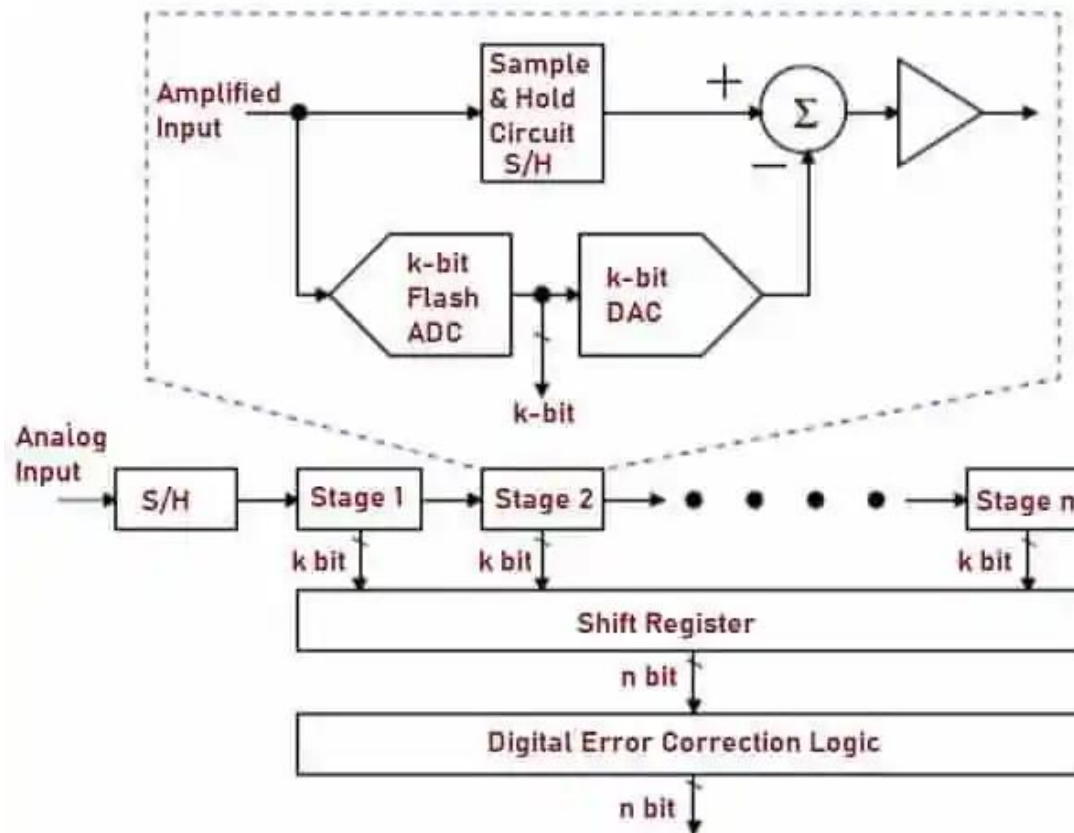
# ADC Types

1. **Flash ADC:** are used in the applications which requires High-Speed and large bandwidth such as *radar detection*. It is also known as **Parallel ADC**.



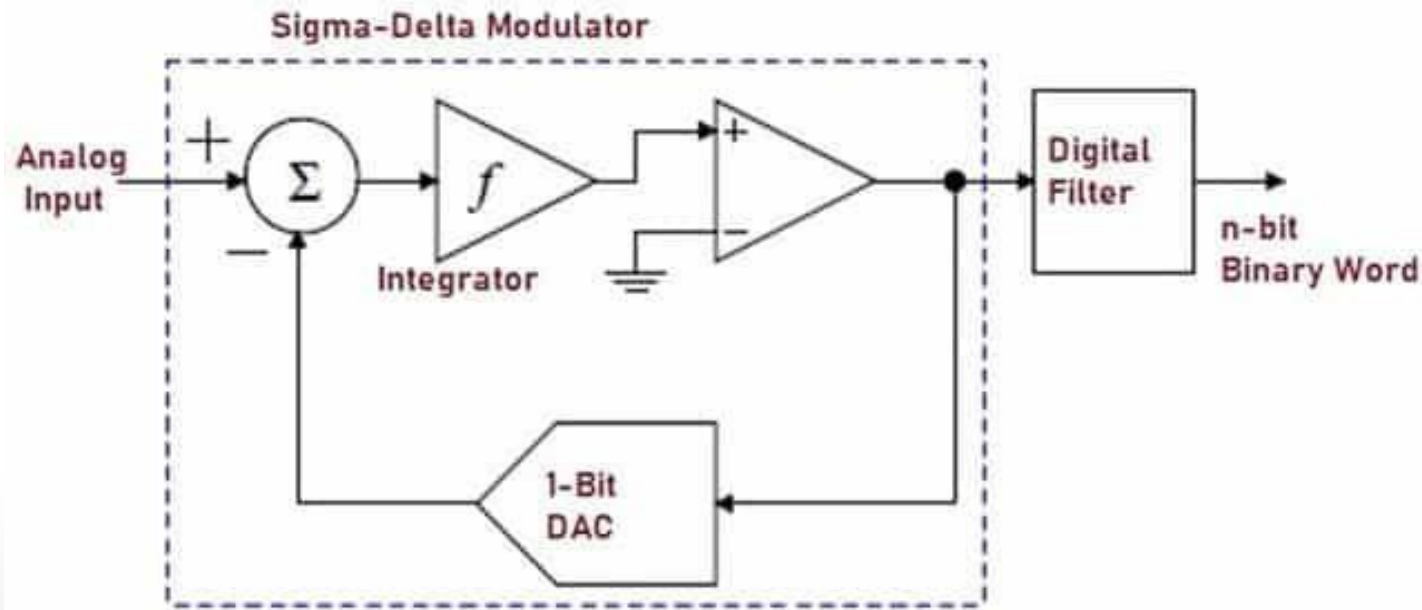
# ADC Types

2. **Pipelined ADC:** are designed using two or more low resolution Flash ADCs. It's mainly used in *digital oscilloscopes, spectrum analyzers*.



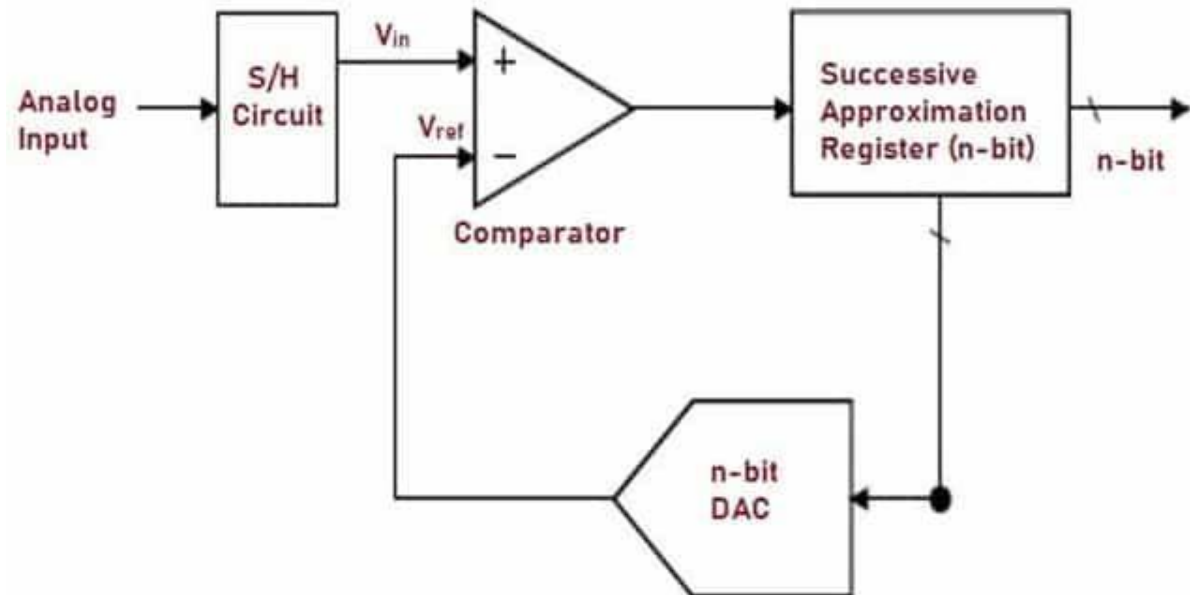
# ADC Types

3. **Sigma Delta ADC:** The architecture of Sigma-Delta ( $\Sigma$ - $\Delta$ ) converters, also called as Oversampling Converters is quite simple. It consists of two main block: **sigma-delta modulator** and **Digital filter**. It's mainly used in *telephone* and *radio communications*.



# ADC Types

4. **Successive Approximation ADC:** This type of Analog to Digital Converter incorporates Successive Approximation Algorithm to convert analog input to a digital binary code. It's mainly used *in CMOS Image sensors for mobile applications.*





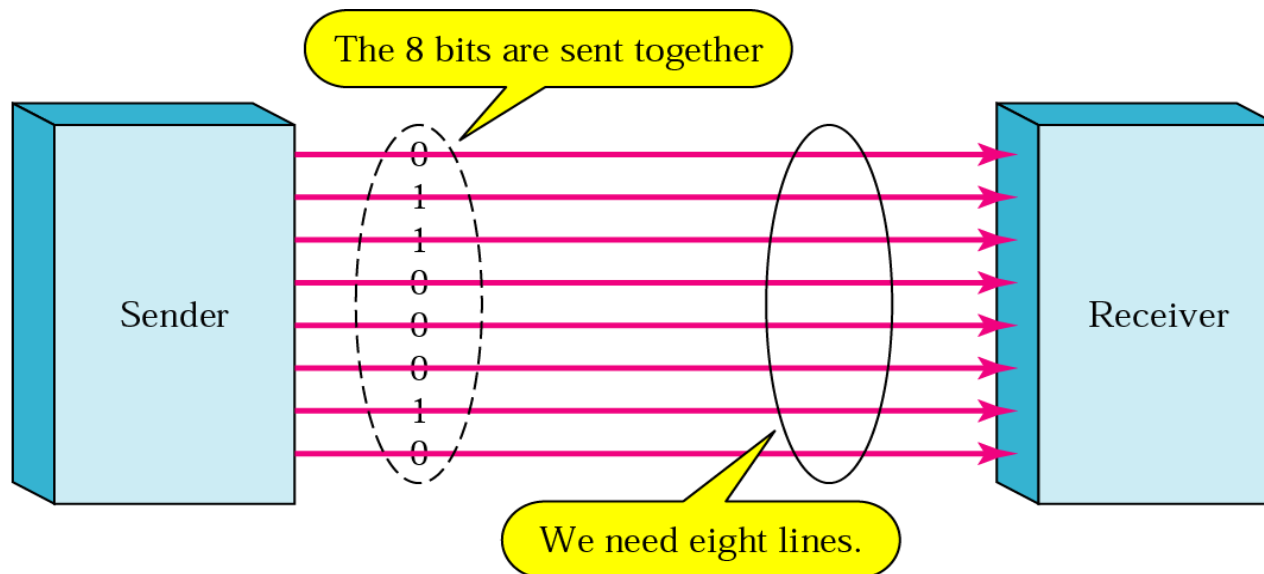
# Transmission Mode

- Parallel Transmission
- Serial Transmission



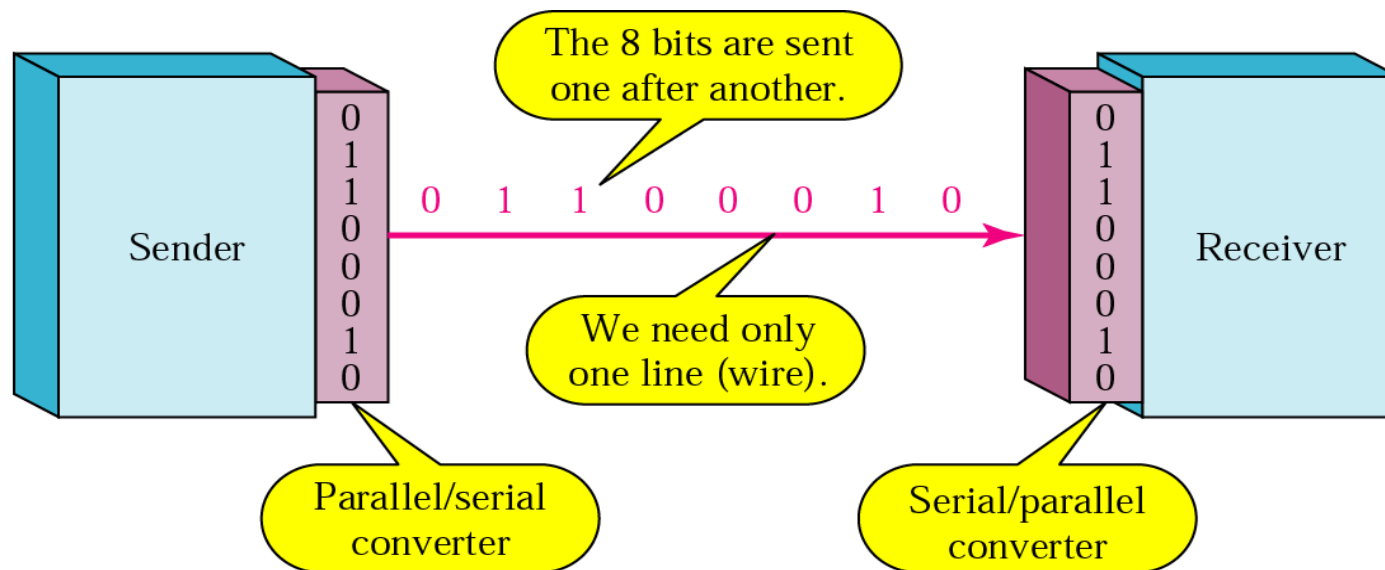
# Transmission Mode

1. **Parallel Transmission:** multiple data bits are transmitted over multiple channels at the same time. This means that data can be sent much faster than using serial transmission mode.



# Transmission Mode

2. **Serial Transmission:** the data bits are organized in a specific order, since they can only be sent one after another. The order of data bit is important because it's dictates how the transmission is organized when it's received.



# Transmission Mode

Description	Serial	Parallel
Number of bits transmitted at one clock pulse	<b>One</b> bit	<b>n</b> bit
Number of line required to transmit n bits	<b>One</b> lines	<b>n</b> lines
Speed of data transfer	Slow	Fast
Cost of Transmission	Low as one line is required	Higher as <b>n</b> lines are required
Application	Long Distance communication between two computer	Short distance communication like computer to printer
Interference	No Crosstalk problem	Crosstalk creates interference between the parallel lines
Bandwidth	The bandwidth of serial wires is much higher	The bandwidth of parallel wires is much lower
Flexibility	Serial interface is more flexible to upgrade, without changing the hardware	Parallel data transfer mechanism rely on hardware resource and hence not flexible to upgrade

*Differences table between Serial and Parallel transmission mode*

# Homework

1. Pro and Cons of Unipolar, polar, and Bipolar
2. Why do we need to do Sampling?
3. Why do we need Quantization?
4. Why do we need Binary Encoding?
5. Pro and Cons of Serial and Parallel Transmission

# Q&A

