


# Introduction to Telecommunication Systems

## Lecture 9



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# Digital Transmission

## Analog Data, Digital Signal

- Two techniques are used for analog to digital conversion:
  - Pulse code modulation and
  - Delta modulation



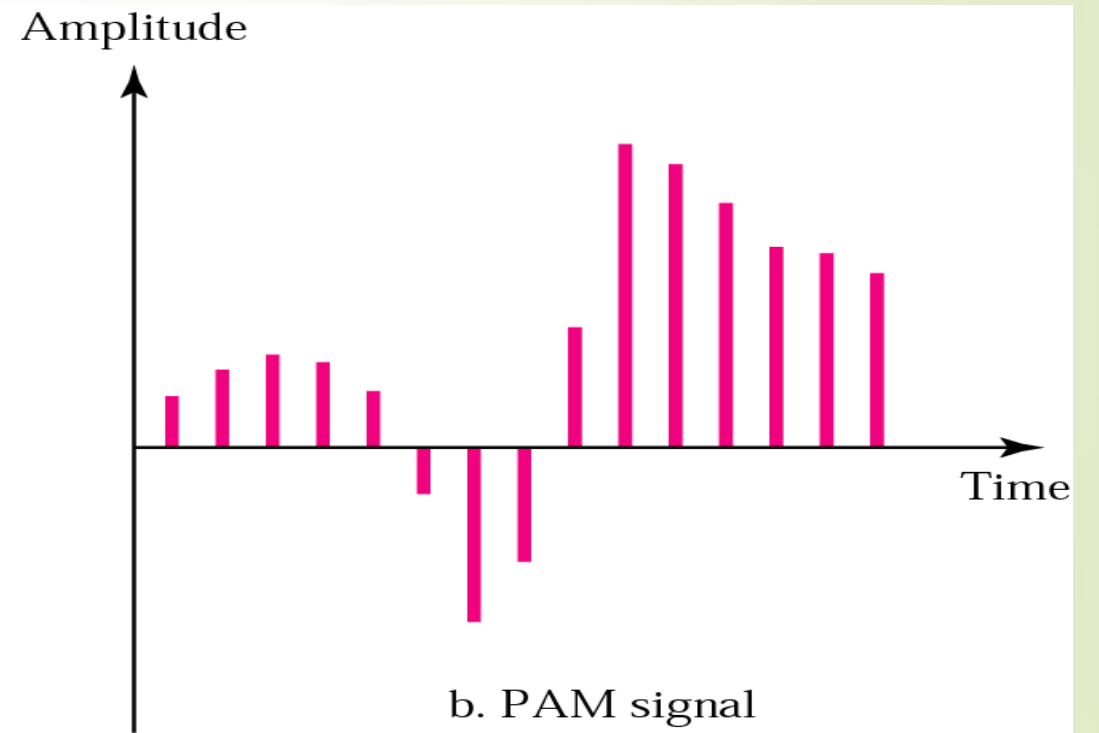
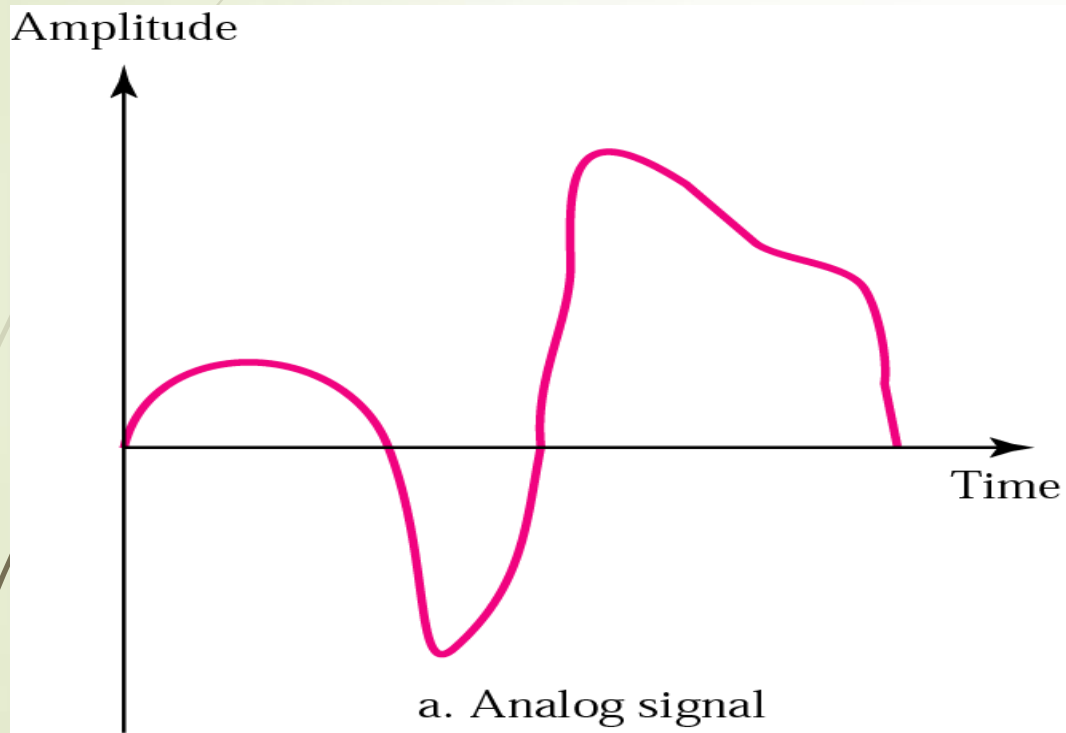
# Pulse Code Modulation

- The most common technique to change an analog signal to digital is called pulse code modulation (PCM).
- A PCM encoder has three processes:
  - The analog signal is sampled.
  - The sampled signal is quantized.
  - The quantized values are encoded as streams of bits.

# Sampling

- The first step in PCM is sampling.
- The sampling process is sometimes referred to as pulse amplitude modulation (PAM).
- The analog signal is sampled every  $T_s$  s, where  $T_s$  is the sample interval or period.
- The inverse of the sampling interval is called the sampling rate or sampling frequency.
- According to Nyquist Theorem, Sampling rate must be at least 2 times the highest frequency.

# Sampling



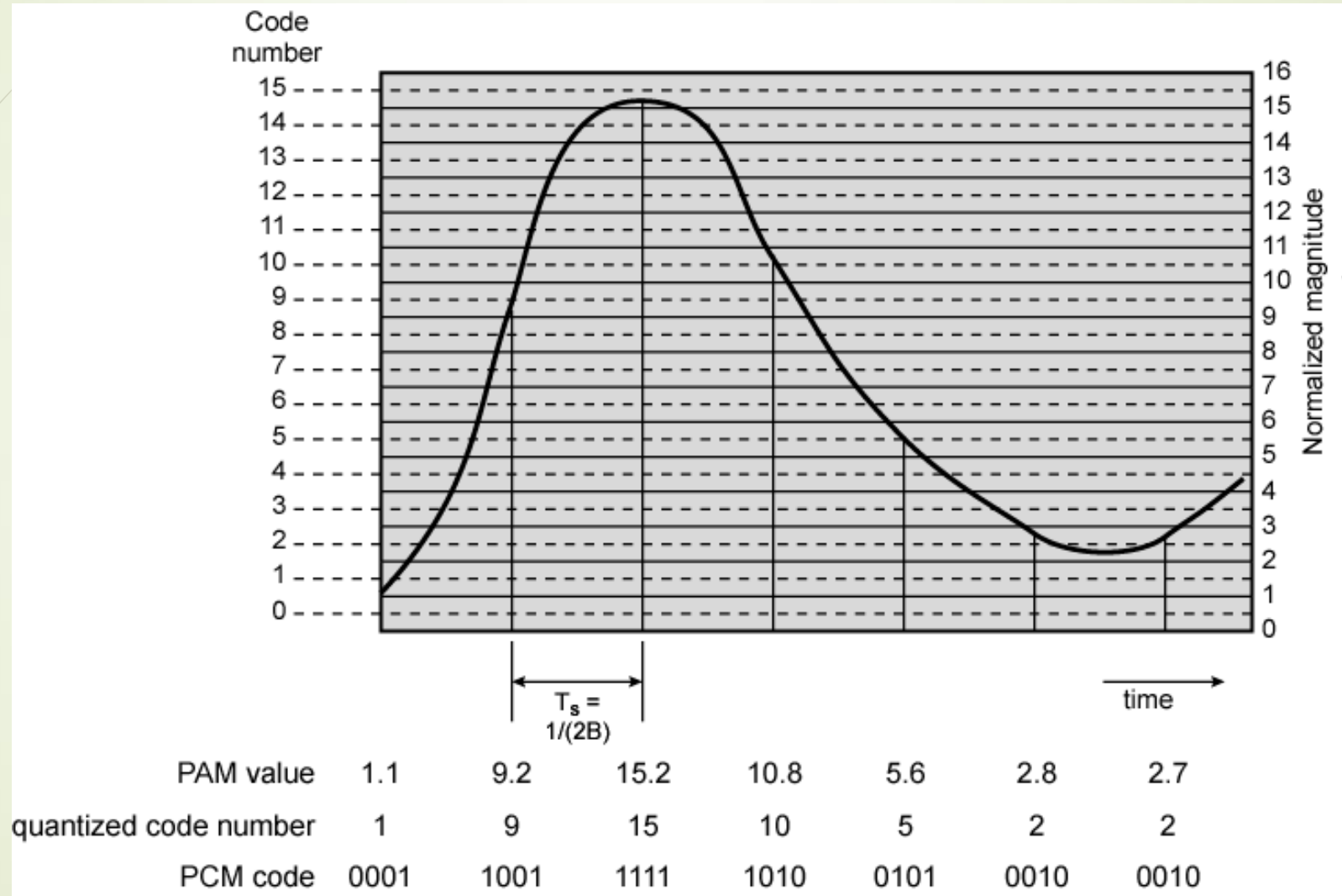
# Quantization

- The result of sampling is a series of pulses with amplitude values between the maximum and minimum amplitudes of the signal.
- The set of amplitudes can be infinite with non integral values between the two limits.
- These values cannot be used in the encoding process.
- To convert to digital, each analog sample must be assigned a binary code.

# Encoding

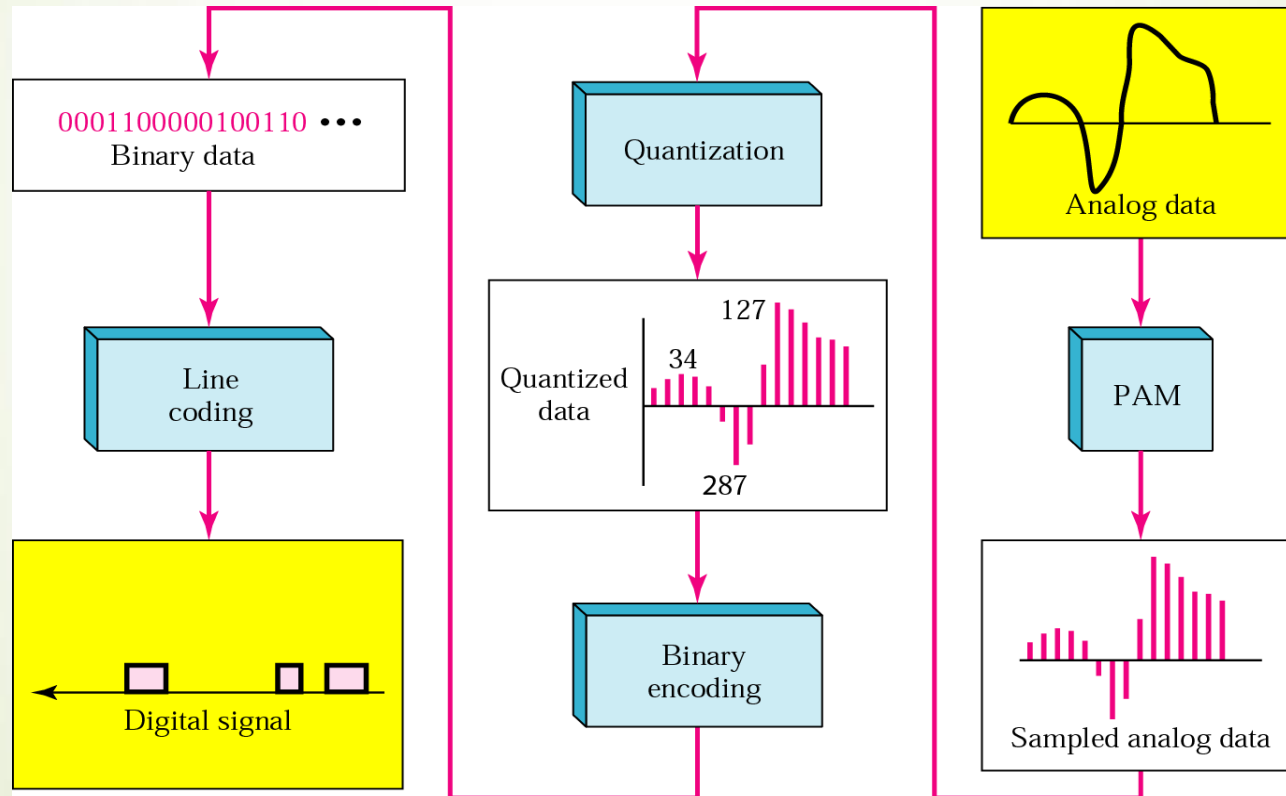
- The last step in PCM is encoding.
- After each sample is quantized and the number of bits per sample is decided, each sample is assigned a binary code.
- Note that the number of bits for each sample is determined from the number of quantization levels.
- For example, 8 bit sample gives 256 levels.
- 8000 samples per second and 8 bits per sample gives 64kbps, for a single voice signal.

# Pulse Code Modulation





# Pulse Code Modulation



# Pulse Code Modulation

- Examples

- Q1: What sampling rate is needed for a signal with a bandwidth of 10 KHz (1KHz to 11KHz)

- A1: Sampling rate =  $2 \times 11 \text{ KHz} = 22,000$  samples per second

# Pulse Code Modulation

- Examples

- Q2: A signal is sampled. Each sample requires at least 12 levels of precision (+0 to +5 and 0 to -5). How many bits should be sent for each sample?

- A2: 4-bit

- 1-bit for sign

- 3-bit for magnitude (8-levels)

# Pulse Code Modulation

- Examples

- Q3: We want to digitize the human voice. What is the bit rate, assuming 8-bits per sample?

- A3: BW of Human voice 0-4000 Hz

- Sampling rate  $4000 \times 2 = 8000$  samples/sec

- Bit rate

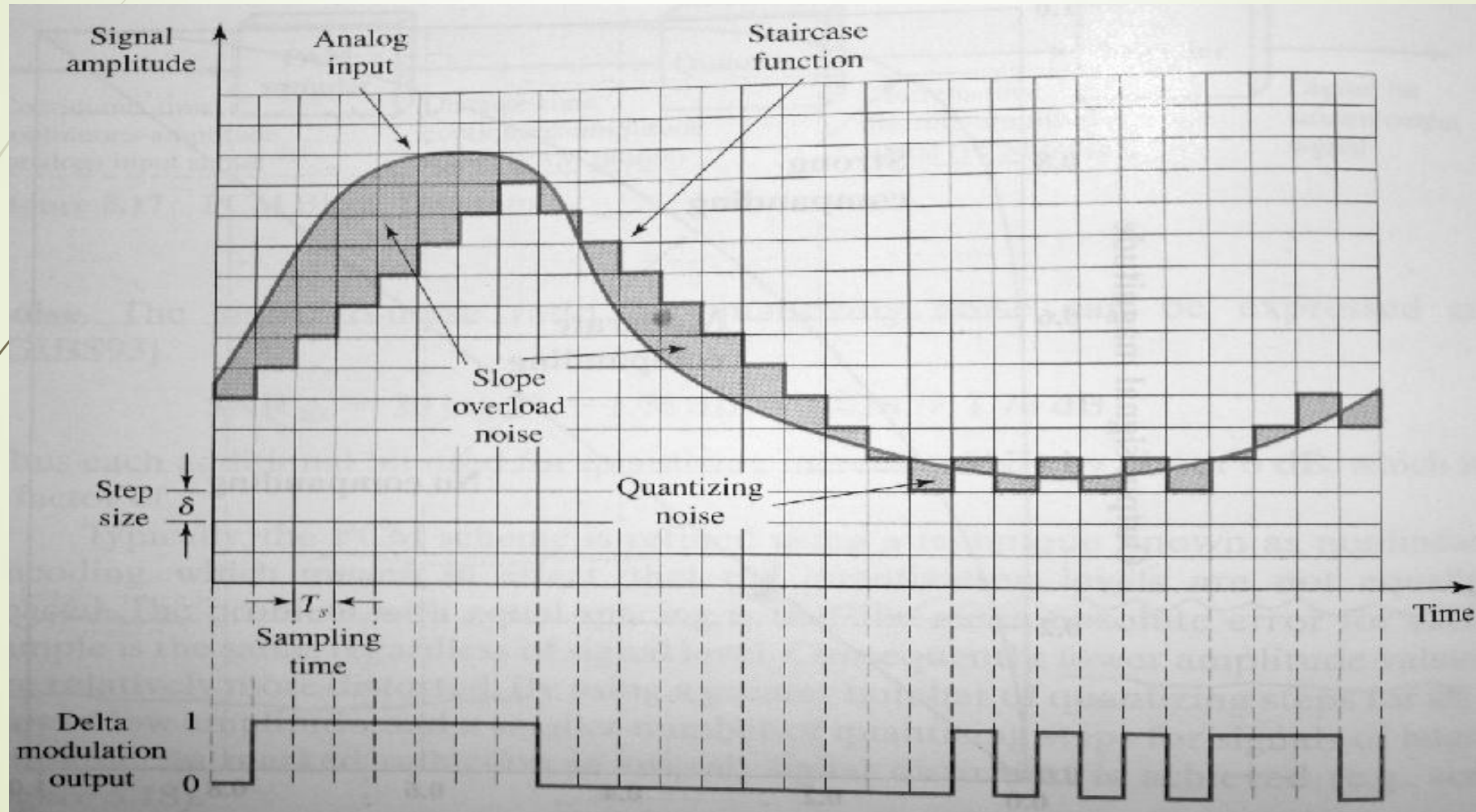
- $8000 \text{ sample/sec} \times 8 \text{ bit/sample}$   
 $= 64,000 \text{ bps}$



# Delta Modulation

- Modulation:
  - An analog signal is approximated by a staircase function that moves up or down by quantization level at each sampling interval.
  - If the value of the sampled waveform exceeds that of the staircase function, 1 is generated, otherwise, 0 is generated.
- Two important parameters:
  - The size of the step.
  - The sampling rate.

# Delta Modulation



# Delta Modulation

- Noise
  - Slope overload noise (when the analog waveform is changing rapidly than the staircase can follow)
  - Quantizing noise (when the analog waveform is changing slowly)
- Trade-off
  - The quantizing noise increases as the size of the step increases.
  - The slope overload noise increases as the size of the step decreases.



End of Slides