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**SAIRAM**  
DIGITAL RESOURCES

## UNIT 2 PULSE AND DATA COMMUNICATION

- 2.1.3 PULSE CODE MODULATION



**EC8394**

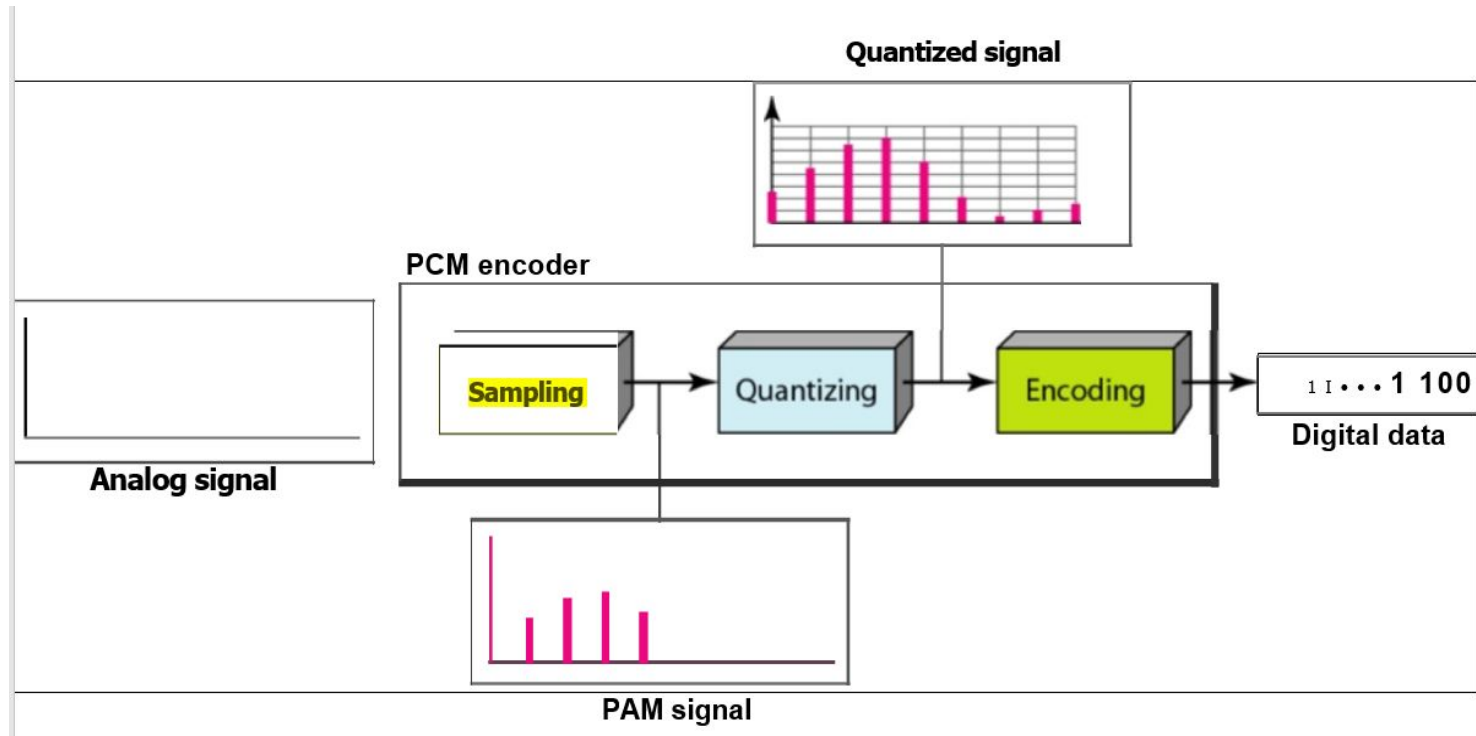
**ANALOG AND DIGITAL COMMUNICATION**

**ELECTRONICS & INSTRUMENTATION ENGINEERING**



- **Introduction**
- **Block diagram of PCM**
- **PCM processes**
- **PCM standards**
- **Bit rate and bandwidth requirements of PCM**
- **Advantages**
- **Disadvantages**
- **Application**

- Analog voice data must be translated into a series of binary digits before they can be transmitted.
- With Pulse Code Modulation (PCM), the amplitude of the sound wave is sampled at regular intervals and translated into a binary number.
- The difference between the original analog signal and the translated digital signal is called *quantizing error*.



# Filtering

# Sampling

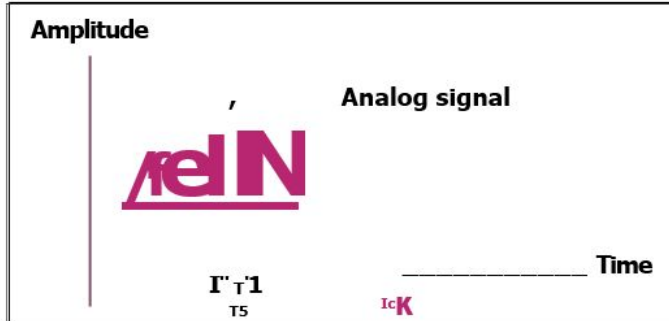
# Quantization

# Encoding

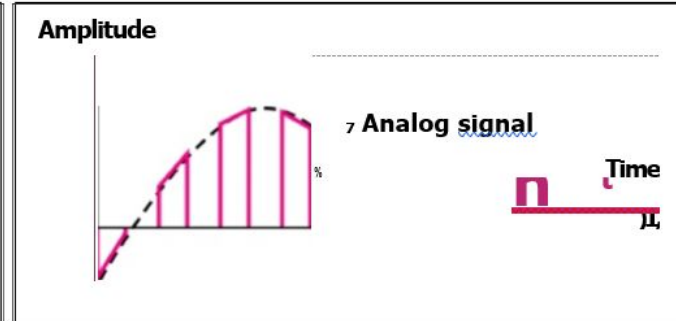
- Analog signal is sampled every  $T_s$  sec.
- $T$  is referred to as the sampling interval.
- $1/T_s$  called the sampling rate or sampling frequency.

**There are 3 sampling methods:**

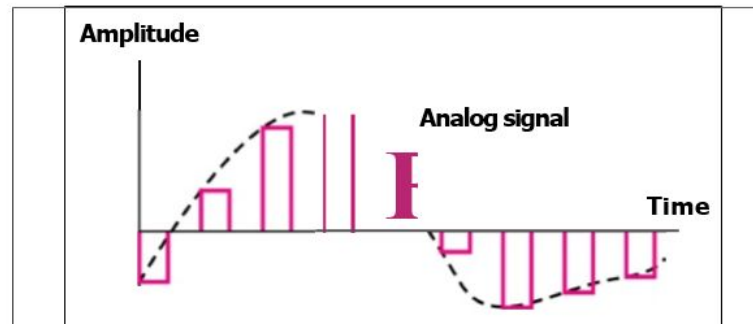
- Ideal - an impulse at each sampling instant
- Natural - a pulse of short width with varying amplitude
- Flat top - sample and hold, like natural but with single amplitude value



a. Ideal sampling



b. Natural sampling



c. Flat-top sampling



- The process of measuring the numerical values of
- the samples and giving them a table value in a suitable scale
- D The finite number of amplitude intervals is called the quantizing interval' like quantizing interval no.1 is 10-20mV; 2 is 20-30mV etc. in a case of 1V signal.
- Linear quantizing is where the quantizing intervals are of the same size



- Quantization intervals are coded in binary form, and so the quantization intervals will be in powers of 2.
- In PCM, 8 bit code is used and so we have 256 intervals for quantizing (128 levels in the
- Positive direction and 128 levels in negative direction)

- The deviation between the amplitude of samples at the transmitter and receiving ends
- In linear quantization, the distortion is more and to decrease the distortion, the no. of steps in the given amplitude range has to be increased.
- Due to BW limitations, more quantum levels in small amplitude region are planned results to
- Non linear (uniform) quantization

- There are two standards of pcm namely
  - 1) The European Standard
  - 2 ) The American Standard
- They differ slightly in the detail of their working but the principles are the same.

European pcm = 30 channels

North American pcm = 24 channels

Japanese pcm = 24 channels

In India we follow the European pcm of 30 channels system working.

- **Uniform Transmission Quality Compatibility of different classes of Traffic in the Network**
- **Integrated Digital Network increased utilization of Existing Circuit**
- **Low Manufacturing Cost**
- **Good Performance Over Very poor Transmission Paths**

- ❑ **Large Bandwidth required for Transmission**
- ❑ **Noise and crosstalk leaves low but rises attenuation**
- ❑ **An integrated Digital network can only be realized be a gradual extension of Noise**

<https://www.youtube.com/watch?v=HIGJ6xxbz8s>