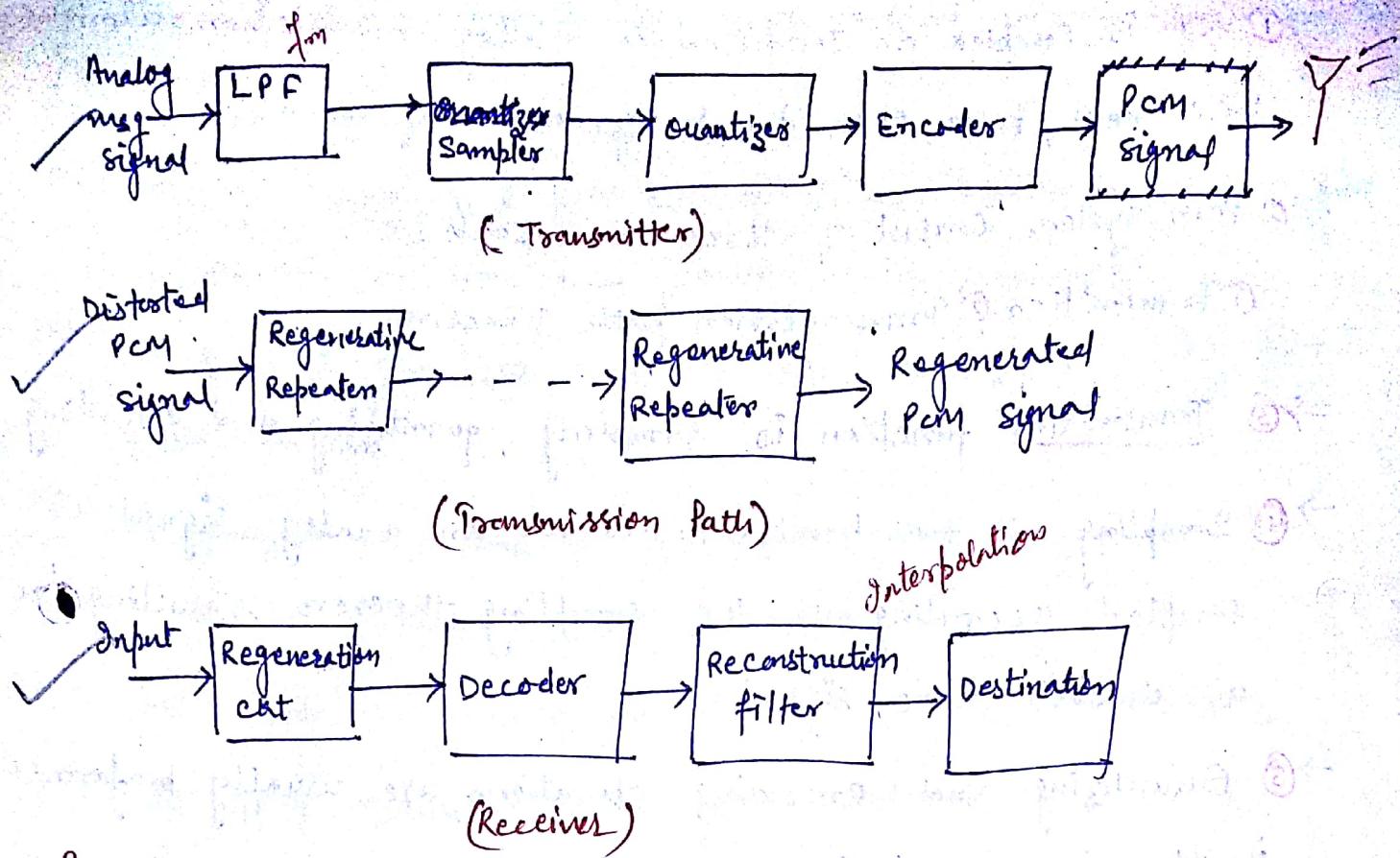
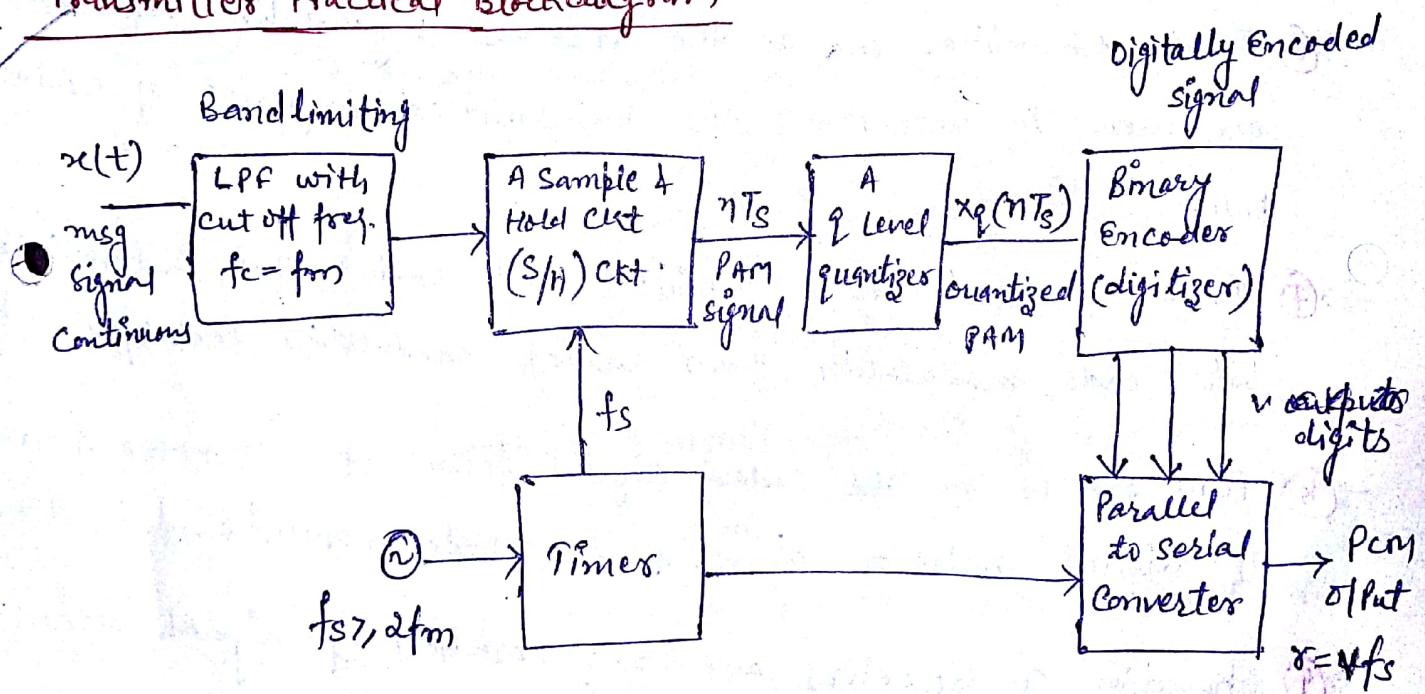


# Basic Block Diagrams of PCM



## ① Transmitter Practical Block Diagram



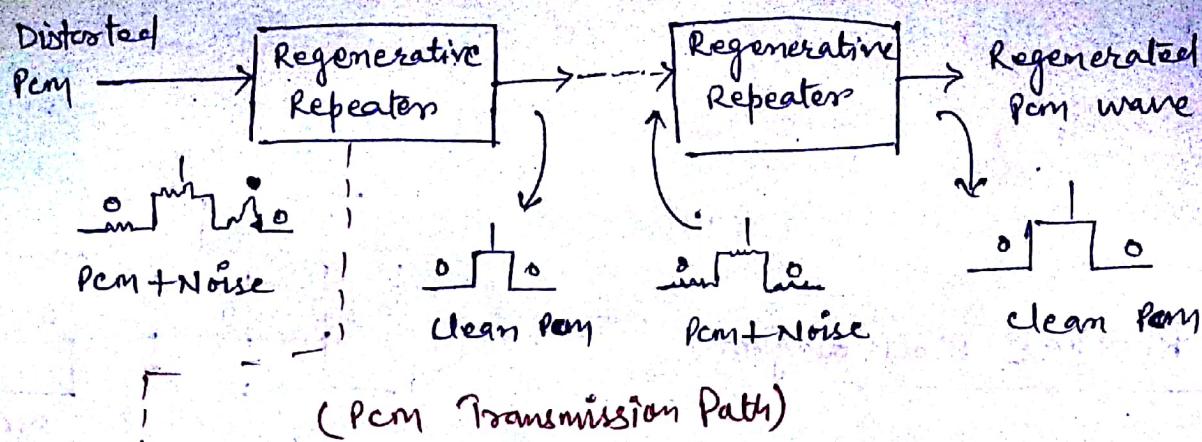
- ① signal  $x(t)$  first passed through the low pass filter of cut off freq.  $f_m$  Hz. This LPF blocks all frequency above  $f_m$  Hz.
- ② Now the  $x(t)$  is band limited to  $f_m$  Hz.

## PCM (Pulse Code Modulation)

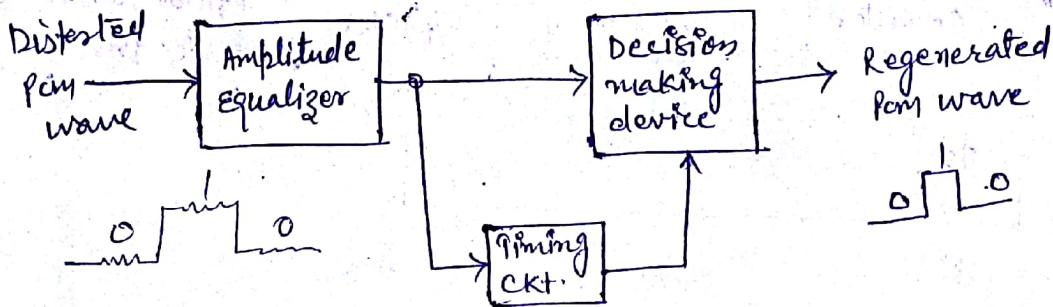
- ① PCM is complex as compared to analog modulation technique.  
Like PAM, PAM, PPM. due to great no. of operations.
- ② PCM systems consist of three main parts i.e.  
① Transmitter, ② Transmission path, ③ Receiver.
- ③ Transmitter function is sampling, quantizing and encoding.
- ④ Sampling is the process in which an analog signal is sampled according to the sampling theorem resulting in a discrete time signal.
- ⑤ Quantizing and encoding operations are usually performed in the same ckt known as ADC.
- ⑥ From transmitter end to the receiver regenerative repeaters are used to reconstruct the transmitted sequence of coded pulse.
- ⑦ It is combined use of quantizing and coding that distinguishes pulse code modulation from analog modulation technique.
- ⑧ PCM opp is in the coded digital form. It is in the form of digital pulse of constant amplitude, width and position.
- ⑨ Operation of receiver are regeneration of signal, decoding and demodulation of the trains of quantized sample.
- ⑩ These operation is performed in the same ckt which is known as DAC.

- (3) Sample and hold ckt then sample this signal at the rate of  $f_s$
- (4)  $f_s$  is selected sufficiently above nyquist rate to avoid aliasing.
- $f_s > 2f_m$
- (5) O/P of sample and hold ckt is denoted by  $x(nT_s)$ . This signal is discrete in time and continuous in amplitude.
- (6) A  $q$  level quantizer compares input  $x(nT_s)$  with its fixed digital levels.
- (7) It assigns any one of the digital level to  $x(nT_s)$  which results in minimum distortion or error.
- (8) This error is quantization error.
- (9) Thus O/P of quantizer is a digital level called  $x_q(nT_s)$ .
- (10) Now the quantized signal level  $x_q(nT_s)$  is given to binary encoder. This encoder converts input signal  $v$  digits binary word.
- (11) Thus  $x_q(nT_s)$  is converted to ' $v$ ' binary bits.
- (12) This encoder is known as digitizer.
- Note: (1) It is not possible to transmit each bit of the binary word separately on transmission line.
- (2) Therefore  $v$  binary bits (digits) are converted to serial bit stream to generate single base band signal.
- (3) Parallel to serial conversion shift register does this job.
- (4) O/P of PCM Generator is thus a single base band signal of binary bits.

- ⑤ An <sup>Timer</sup> oscillator generates the clocks for sample and hold circuit and parallel to serial converter.
- ⑥ In the PCM sample and hold, quantizer and encoder combinedly form an analog to digital converter (ADC).
- ② PCM Transmission Path
- ① Path over which PCM signal travels b/w PCM Transmitter and PCM receiver is called PCM Tr. path.
- ② Problem in PCM systems lies in its ability to control the effects of distortion and noise when the PCM wave travels on the channels.
- ③ PCM accomplishes this capacity by means of using a chain of regenerative repeaters.
- ④ Repeaters are spaced close enough to each other on the transmission path.
- ⑤ Regenerative performs basic three operations Equalization, timing and decision making.
- ⑥ Hence each repeater actually reproduces the clean noise free PCM signal from the PCM signal distorted by the channel noise.
- ⑦ This improves the performance of PCM in presence of noise.



Regenerative Repeaters Block diagram

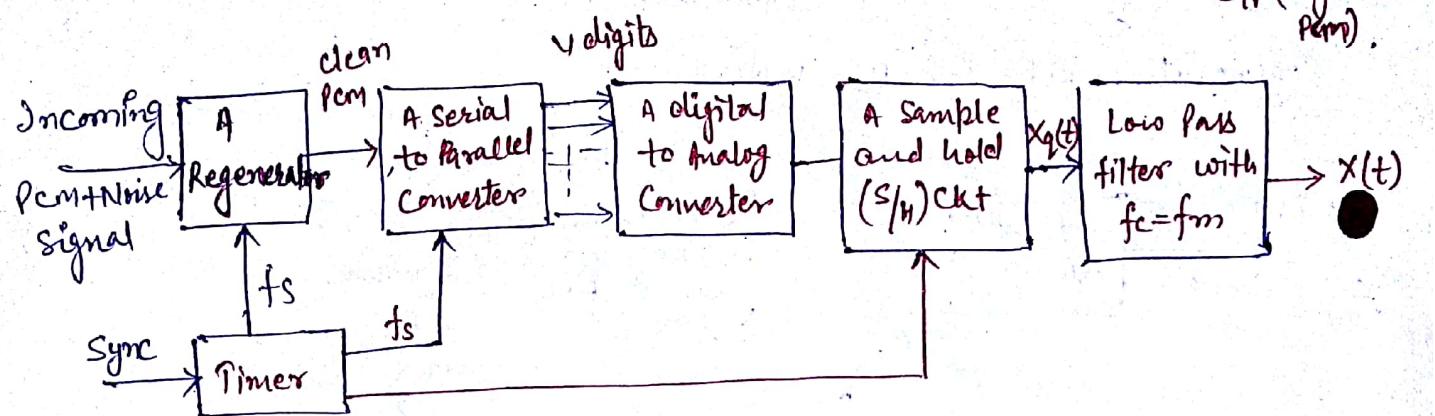


- ① Amplitude equalizer shapes the distorted Pcm wave so as to compensate for the effects of amplitude and phase distortions.
- ② Timing ckt. produces a periodic pulse train which is derived from the input Pcm Pulses.
- ③ Pulse train is then applied to the decision making device.
- ④ The decision making devices uses this pulse train for sampling the equalized Pcm Pulses.
- ⑤ The sampling is carried out at the instants where the signal to noise ratio is maximum.
- ⑥ The decision device makes a decision about whether the equalized Pcm wave at its input has a 0 value or 1 value at the instant of sampling.

⑦ Such a decision is made by comparing equalized Pcm with a reference level called decision threshold.

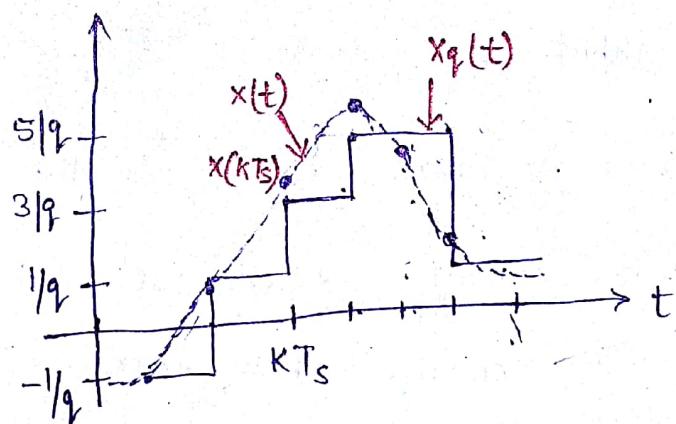
⑧ At the output of decision device we get a clean Pcm signal without any trace of noise.

### PCM Receiver Block diagram



① Regenerator at the start of PCM receiver reshapes the pulse and removes the noise.

② This signal is then converted to parallel digital words for each sample.



③ Now, the digital word is converted to its analog value denoted as  $x_q(t)$  with the help of a sample and hold ckt.

④ This signal, at the output of sample and hold ckt, is allowed to pass through a low pass reconstruction filter to get the appropriate original msg signal denoted as  $x(t)$ .

- ⑤ It is impossible to reconstruct exact original signal  $x(t)$  & bcz of permanent quantization error introduced during quantization at the transmitter.
- ⑥ This quantization error can be reduced by <sup>(4)</sup> increasing the binary levels.
- ⑦ This is equivalent to taking binary digits per sample.
- ⑧ But taking bits ' $n$ ' taking the signalling rate as well as transmission bandwidth as we have already done.  
Therefore noise due to quantization error is <sup>in</sup> tolerable limits.

## Applications of Pcm

- ① with advent of optical cable Pcm is used in telephony
- ② In space comm. transmit power is very low (10 to 15W) and distance is very large, due to high noise immunity only Pcm systems can be used in such application.

## Advantages of Pcm

1. High noise immunity
2. Due to digital ~~is~~ nature we can place repeaters b/w to and ~~repeater~~ receiver to reduce the effect of noise.
3. Storage possibility is more.
4. Using different Coding technique only desired person can decode the received signal.

## Drawbacks of Pcm

1. The encoding, decoding and quantizing circuitry of Pcm is complex
2. Pcm requires a large Bandwidth as compared to the other system

→ Digital level are chosen as  $\pm \frac{D}{2}$ ,  $\pm \frac{3D}{2}$  etc to reduce the quantization noise or error.

### Transmission Band width in a PCM Systems →

⇒ Let us assume that quantizer use 'v' no. of binary digits to represent each level.

⇒ Then no. of levels that may be represented by v digits will be.

$$q = 2^v$$

Total no. of digital level of a q level quantizer.

For Example. if  $v=4$  bits then total No. of level will be.

$$q = 2^4 = \underline{\underline{16 \text{ levels}}}$$

Each sample is converted to v binary bits ie

$$\text{No. of bits per sample} = v.$$

we know that

$$\text{No. of sample per second} = f_s$$

⇒ Therefore No. of bits per second is  $= \left[ \begin{array}{l} \text{No. of bits per samples} \times \text{No.} \\ \text{(x)} \end{array} \right] \text{ of sample per second}$

⇒ No. of bits per second is known signaling rate of PCM.

$$\checkmark [r = v f_s]$$

$$\text{where } f_s \geq 2f_m$$

Therefore:

$$BW \geq \frac{1}{2} r$$

$$r = v f_s \quad \text{or} \quad r = v \cdot 2f_m.$$

Note: Since BW needed for PCM transmission is half of the signaling rate.

$$\begin{aligned} BW &= \frac{1}{2} r \\ &= \frac{1}{2} \times v \cdot 2f_m \\ \boxed{BW} &= \underline{\underline{v f_m}} \end{aligned}$$

- Note:
- ① In input off char. of a uniform quantizer of the midtread type, which is so called because the origin lies in the middle of a tread of the staircase like graph.
  - ② In midrise type origin lies in the middle of a rising part of the staircase like graph.
  - ③ Both types of uniform quantizers midtread and midrise are symmetric about the origin.

(\*) Signal to quantization noise ratio for PCM systems

$$\frac{S}{N} = \frac{\text{Signal Power}}{\text{Noise Power}} = \frac{P}{\frac{D^2}{12}}$$

$$\frac{S}{N} = \frac{P}{\left(\frac{2x_{\max}}{2^V}\right)^2} \times \frac{1}{12}$$

$$\frac{S}{N} = \frac{P}{\frac{x_{\max}^2}{2^V}} \times \frac{1}{12} \quad \boxed{3}$$

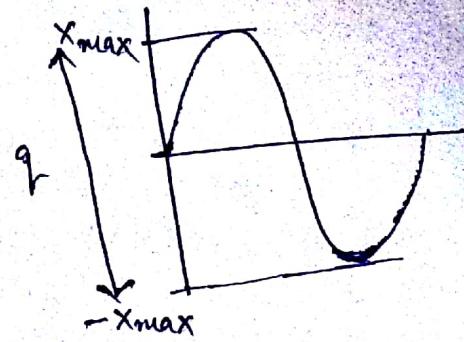
$$\frac{S}{N} = \frac{P \times 3 \times 2^V}{x_{\max}^2}$$

$$\frac{S}{N} = -3 \times 2^{2V}$$

In dB

$$10 \log_{10} \left( \frac{S}{N} \right) \text{dB} = 10 \log_{10} (3 \times 2^{2V})$$

$$\left[ \left( \frac{S}{N} \right) \text{dB} \leq (4.8 + 6V) \text{dB} \right] \checkmark$$



$$\begin{aligned} \text{Total Amplitude} &= x_{\max} - (-x_{\max}) \\ &= 2x_{\max} \end{aligned}$$

$$\boxed{D = \frac{2x_{\max}}{q_r}}$$

or

$$\boxed{q_r = 2^V}$$

$$\begin{cases} x_{\max} = 1 & \text{Normalized input } x(t) \\ P \leq 1 & \text{Normalized Power } P \end{cases}$$