

Question - 1

Q1 List the advantage of SSB Modulation Scheme.

- Answer
- 1) As only one sideband is transmitted, there is a further reduction in transmitter power.
  - 2) As the carrier is not transmitted, this enables 50% reduction in transmitted power level.
  - 3) Receiver bandwidth can be reduced by half which improves S/I to noise ratio by a factor of two.

Question - 2

Q2 Write the power relation for AM signal.

$$P_c = \frac{Ae^2}{2R}$$

$$P_{SSB} = P_{LSB} = \frac{Ae^2 \mu^2}{8R}$$

$$P_t = P_c + P_{SSB} + P_{LSB}$$

$$P_t = \frac{Ae^2}{2R} \left( 1 + \frac{\mu^2}{2} \right)$$

$$P_t = Pe \left( 1 + \frac{\mu^2}{2} \right)$$

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### Question(3)

Ques(3)

Explain deviation ratio for FM Signal -

FM deviation ratio can be defined as the ratio of the maximum carrier frequency deviation to the highest audio modulating frequency.

$$m = \frac{\text{max. freq. Deviation}}{\text{no. of modulation freq.}}$$

Ques(4)

Ques 4

What is basic difference in Direct and Indirect method.

of FM Generation:

A1. Direct method

The carrier frequency  $P_c$  directly varied in accordance with the QPSK baseband sig.

Indirect Method

The modulating sig is used to produce a NB FM sig & frequency multiplication is next used to increase the freq. deviation to the desired level.

### Question 5

Or Define PSD. What is PSD of AWGN?

→ PSD is power spectral density at the sig describe the power present in the signal as a fcn of a freq per unit freq.  
It is expressed  $P_n$  watt/Hz.

- Thermal noise is described by zero mean gaussian random process n(t) that adds onto sig  $\Rightarrow$  additive noise
- If PSD is flat,  $P_n$  is called white noise

Question ⑥ write the condition for threshold in FM modulating scheme

FM threshold  $P_s$  usually defined as a carrier to noise ratio at which demodulating signal-to-noise falls 1dB below the linear relationship. Condition for threshold.

In fm modulating scheme :-

$$f_{OM} = \frac{SNR o/p}{SNR i/p}; f_{OM} > 1$$

$$\boxed{B > \sqrt{\frac{2}{3}}}$$

Question ⑦

Question ⑦ what are the factors affecting PCM transmission bw?

The transmission bw of PCM is -

$$B \cdot w \geq \pi/2 \quad (\pi = \text{sampling rate})$$

$$\pi = 2fs$$

$$B \cdot w \geq \frac{Vfs}{2}$$

$$B \cdot w \geq Vf_m$$

It depends upon -

1) frequency of msg sig.

2)  $\pi$  i.e. no: of bits.

Question ⑧

Ques what is Sampling? Name different type of Sampling  
Sampling is defined as, "The process of measuring the instantaneous values of continuous time signal, in a discrete form". The digitization of sig is called sampling.

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Diffr. Sampling methods are as follows -

- (1) Simple Random Sampling
- (2) Systematic Sampling
- (3) Stratified Sampling
- (4) Clustered Sampling

### Question (9)

Q9. Explain Line Coding with its different types.

The waveform pattern of voltage or current used to represent the Is and Us of a digital sig on a transmission link is called Line Coding.

The common types of Line Coding are -

- (1) Unipolar
- (2) Polar
- (3) Bipolar
- (4) Manchester.

### Question (10)

Q10. What modulation technique is best among ASK, FSK & PSK? Support your answer with proper explanation?

PSK is the best modulation technique because of the following reasons.

- (1) It has a constant envelope.
- (2) In PSK, there is less chance in changing the phase of the sig.

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## Section - B

### Question - 12.

Q Derive the expression of power content  $P_m$  in an AM wave when modulating sig is a single tone sinusoidal sig.

In Single tone amplitude modulation

$$S_{AM}(t) = A_c \{ 1 + K_a m(t) \} \cos 2\pi f_c t$$

$$\text{Let } m(t) = A_m \cos 2\pi f_m t$$

$$S_{AM}(t) = A_c \{ 1 + K_a A_m \cos 2\pi f_m t \} \cos 2\pi f_c t$$

where,  $K_a A_m = f_L$  (modulation index)

$$S_{AM}(t) = A_c \{ 1 + f_L \cos 2\pi f_m t \} \cos 2\pi f_c t$$

$$\begin{aligned} &= A_c \cos 2\pi f_c t + A_c f_L \cos 2\pi f_m t \cdot \cos 2\pi f_c t \\ &= A_c \cos 2\pi f_c t + \frac{A_c f_L}{2} \left\{ \cos 2\pi (f_c + f_m)t + \cos 2\pi (f_c - f_m)t \right\} \\ &\quad \downarrow \quad \downarrow \quad \downarrow \\ &\text{Carrier} \quad \text{VS B} \quad \text{LSB} \end{aligned}$$

Total power,  $P_t = P_c + P_{USB} + P_{LSB}$

$$P_c = \frac{A_c^2}{2R}$$

$$P_{USB} = \left( \frac{A_c f_L}{2} \right)^2 = \frac{A_c^2 f_L^2}{8R}$$

$$P_{LSB} = \frac{A_c^2 f_L^2}{8R}$$

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$$P_{TH} = \frac{A_c^2}{2R} + \frac{A_c^2 \mu^2}{8R} + \frac{A_c^2 \mu^2}{8R}$$

$$= \frac{A_c^2}{2R} \left\{ 1 + \frac{\mu^2}{2} \right\}$$

$$\boxed{P_t = P_c \left\{ 1 + \frac{\mu^2}{2} \right\}}$$

$$\boxed{P_{SB} = \frac{A_c^2 \mu^2}{4R} = \frac{P_c \mu^2}{2}}$$

Efficiency,  $\eta = \frac{P_{SB}}{P_t} \times 100 = \frac{1 + \frac{\mu^2}{2}}{\frac{\mu^2}{2}} \times 100 = \frac{\mu^2 + 2}{\mu^2} \times 100 = \frac{\mu^2 + 2 \times 100}{\mu^2}$

$$\boxed{\eta = \frac{\mu^2 + 2}{\mu^2} \times 100\%}$$

### Question-13

Q- Derive an expression of power content

Q- Derive the mathematical expression for single tone FM -

Ans Carrier sig before modulation

$$c_{u1} = A_c \cos(2\pi f_{ct}t + \phi)$$

$$c_{u1} = A_c \cos(\theta(t))$$

$$\text{where } \theta(t) = 2\pi f_{ct}t + \phi \quad \text{--- (1)}$$

Dif w.r.t t

$$\boxed{\frac{d\theta(t)}{dt} = 2\pi f_c}$$

for Instantaneous frequency,  $f_i$

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$$\theta(t) = \theta_0 + \omega t$$

$$\text{or. } f_i = \frac{1}{2\pi} \frac{d\theta(t)}{dt} \quad \boxed{2}$$

$$\text{we know, } SPM(t) = A_c \cos \theta(t) \quad \boxed{3}$$

From eq. ②

$$\theta(t) = 2\pi \int [f_i dt]$$

$$= 2\pi \int \{f_c + k_f m(t)\} dt$$

$$= 2\pi f_c t + 2\pi k_f \int_{-\infty}^t m(t) dt$$

$$\text{or} \quad = \boxed{2\pi f_c t + 2\pi k_f \int_0^t m(t) dt} \quad \boxed{4}$$

from eq. ③ & ④.

$$SPM(t) = A_c \cos \left\{ 2\pi f_c t + 2\pi k_f \int_0^t m(t) dt \right\}$$

$$\text{let } \underline{m(t)} = A_m \cos 2\pi f_m t$$

$$SPM(t) = A_c \cos \left\{ 2\pi f_c t + 2\pi k_f \frac{A_m \sin 2\pi f_m t}{2\pi f_m} \right\}$$

$$\text{where } \frac{k_f A_m}{f_m} = \beta \text{ (mod. index)}$$

$$\text{Therefore, } \boxed{SPM(t) = A_c \cos \left\{ 2\pi f_c t + \beta \sin 2\pi f_m t \right\}}$$

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### Question (11)

Q11 The antenna current of AM broadcast transmitter modulated to the depth of 40%. by an audio sine wave is 11 Amp. It increases to 12 Amp as a result of simultaneous modulation by another audio sine wave. What is modulation index due to second wave.

$$\text{we know, } P(t) = P_c \left( 1 + \frac{\mu^2}{2} \right) \rightarrow (1)$$

$$\text{Since } P \propto I^2,$$

Eq (1) can be written as

$$\begin{aligned} I_t^2 &= I_c^2 \left( 1 + \frac{\mu^2}{2} \right) \\ \text{or } I_t &= I_c \sqrt{1 + \frac{\mu^2}{2}} \end{aligned} \rightarrow (2)$$

For 1st wave

$$I_t = 11 \text{ A}$$

$$\mu = 0.4$$

$$11 = I_c \sqrt{1 + \frac{0.16}{2}}$$

$$= I_c \times 1.039$$

$$I_c = 10.58 \text{ A}$$

for 2nd wave

$$I_t = 12 \text{ A}$$

$$I_t = P_c \sqrt{1 + \frac{\mu_1^2 + \mu_2^2}{2}}$$

$$12 = 10.58 \sqrt{1 + \frac{(0.16 + \mu_2^2)}{2}}$$

$$1.285 = 1 + \left( \frac{0.16 + \mu_2^2}{2} \right)$$

$$\mu_2 = 0.641$$

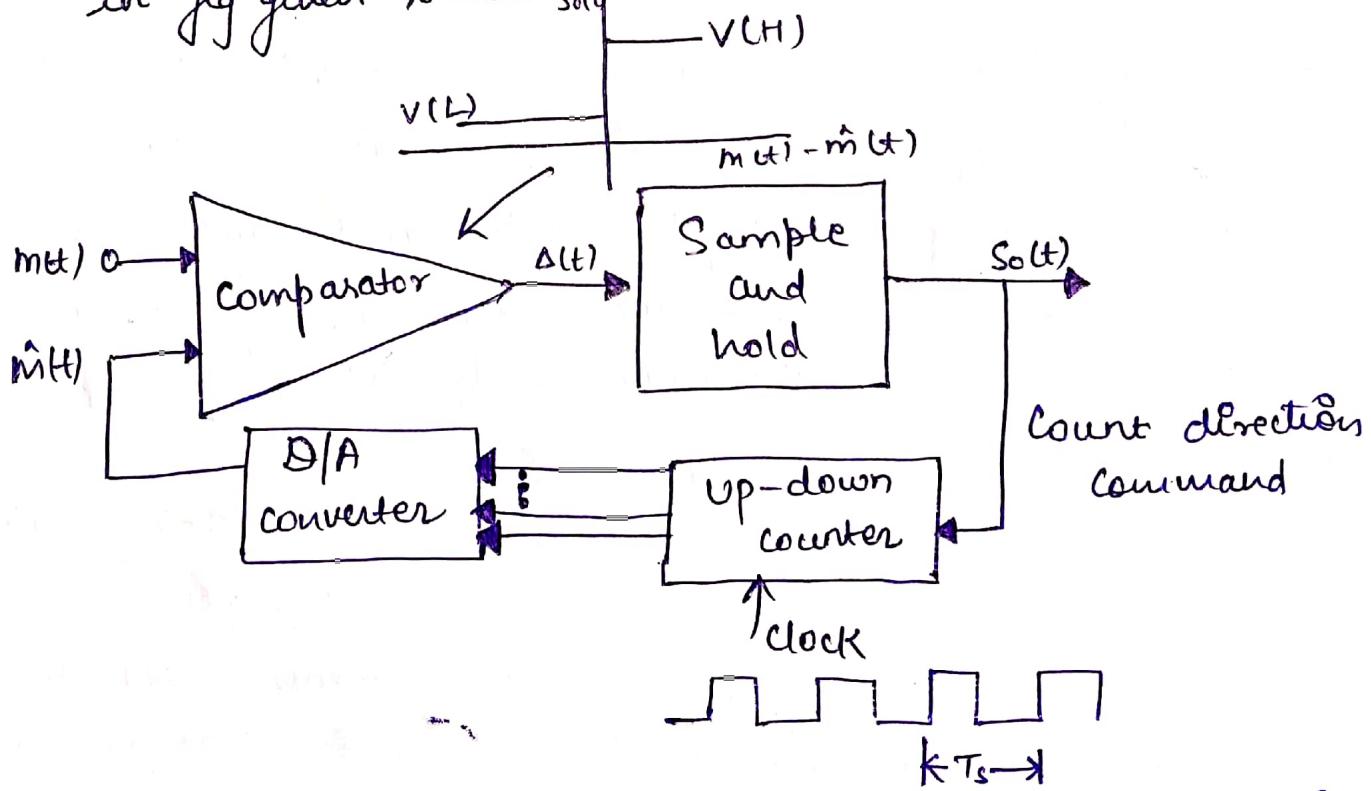
✓  
Solved

Section-C.

Question 16 (b) What is Delta modulation? Discuss the error in DM. How it can be overcome in Adaptive Delta modulation.

Answer. Delta modulation — Delta modulation (DM) is a DPCM scheme in which the difference signal  $\Delta(t)$  is encoded into just a single bit.

- The single bit is used to increase or decrease the estimate  $\hat{m}(t)$ .
- This scheme is called linear delta modulation & is shown in fig given below.



- The baseband sig  $m(t)$  and its quantized approximation  $\hat{m}(t)$  are applied as inputs to a comparator.

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# Comparator - 180273 1079

- A comparator makes a comparison b/w inputs.
- The comparator has 1 fixed o/p  $V(H)$  when  $[ \because m(t) > \hat{m}(t) ]$  and different o/p  $V(L)$  when  $m(t) < \hat{m}(t)$ .

# Up-down Counter - MT XII

- It increments or decrements its count by 1 at each active edge of the clock waveform.

# Counter direction -

- Its incrementing or decrementing, is determined by the voltage levels at the "count direction command" input to the counter.

# The Error in digital modulation

- When the binary input, which is also the transmitted o/p  $s_0(t)$ , is at the level  $V(H)$ , the counter up & when it is at level  $(V(L))$  the counter counts down.

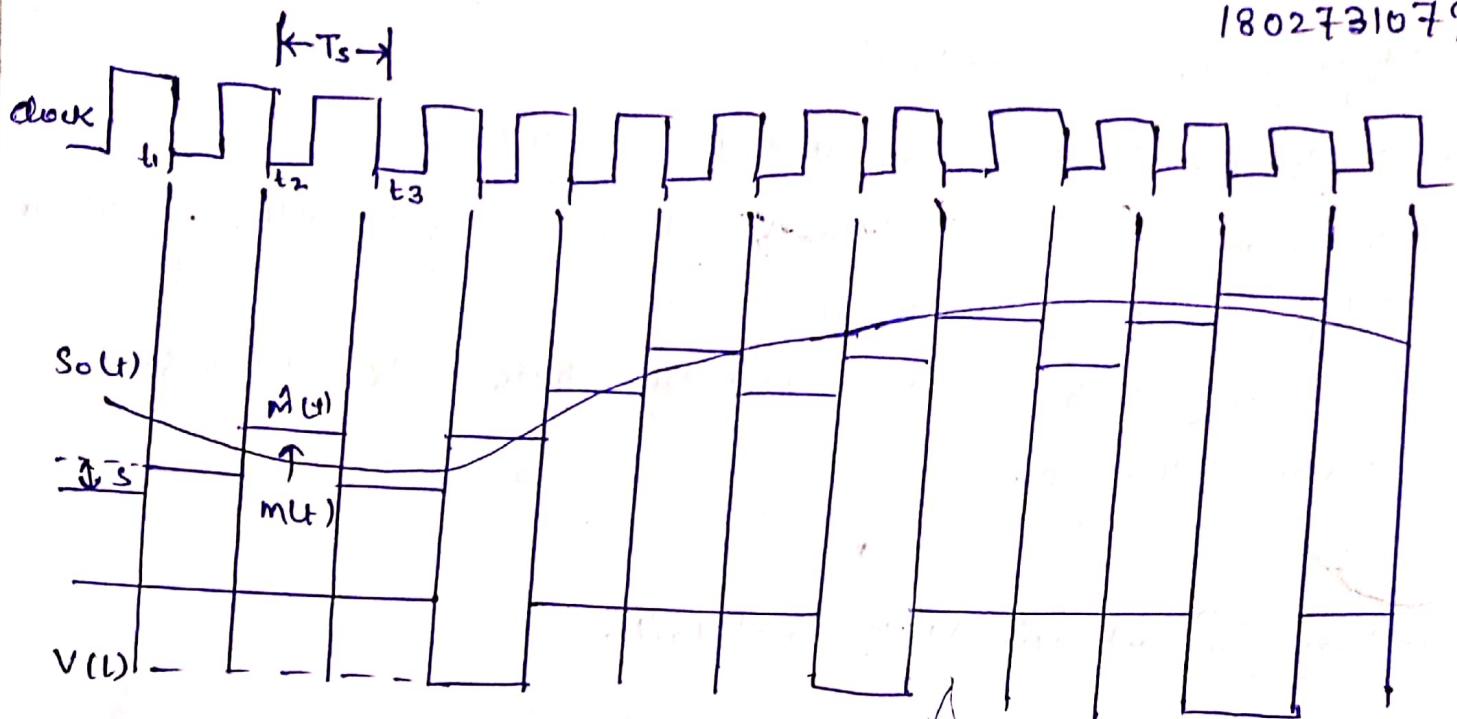
- The digital o/p of the counter is converted to the analog quantized approximation  $\hat{m}(t)$  by the D/A converter.

- For a time preceding  $t_1$ , we find  $m(t) > \hat{m}(t)$  so that  $s_0(t) = V(H)$
- At  $(t_1)$ , when the active clock edge appears, the counter is incremented & immediately s/g  $\hat{m}(t)$  jumps up by an amount equal to the step size  $s$ .

- At  $(t_2)$ , we still find  $m(t) > \hat{m}(t)$ , so  $s_0(t)$  remains at  $V(H)$  and there is another upward jump in  $\hat{m}(t)$ .

- At  $(t_3)$ ,  $m(t) < \hat{m}(t)$ . So,  $s_0(t)$  becomes  $s_0(t) = V(L)$  the counter decrements, & there is a consequent downward jump in  $\hat{m}(t)$  by amount  $s$  and soon.

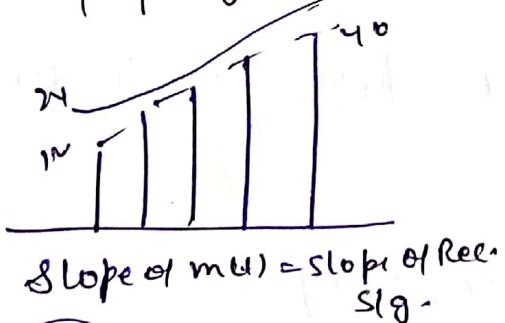
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The excessive disparity b/w  $m(t)$  and  $\hat{m}(t)$  is described as a slope overloaded error and occurs whenever  $m(t)$  has a slope large than the slope  $S/T_s$  which can be sustained by the waveform  $\hat{m}(t)$ .

- Slope overloaded can be avoided if  $\frac{S}{T} > \left| \frac{dm(t)}{dt} \right|_{\max}$ .
- The dynamic range of amplitude of  $m(t)$  is too small because of threshold & overloaded effect.

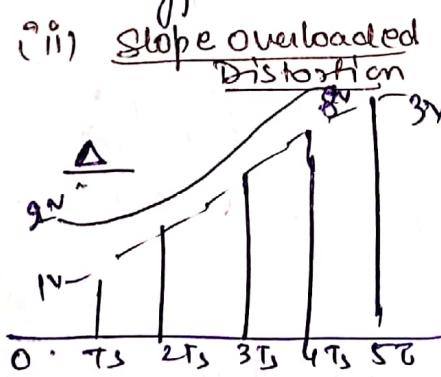
### iii. Perfect Reconstruction



$$\frac{d}{dt} m(t) = \frac{\Delta_{opt}}{T_s}$$

Rate of change of msg slp

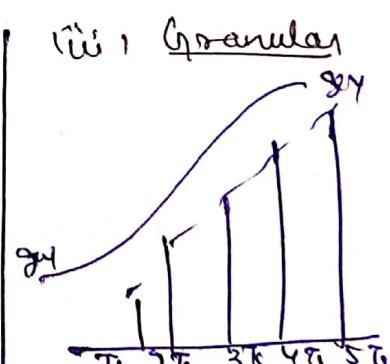
$$\text{Slope} = \frac{\Delta}{T_s}$$



$$\frac{\Delta}{T_s} < \frac{d}{dt} m(t)$$

$$\frac{\Delta}{T_s} < \frac{\Delta_{opt}}{T_s}$$

$$\boxed{\Delta < \Delta_{opt}}$$



$$\frac{\Delta}{T_s} > \frac{d}{dt} m(t)$$

$$\frac{\Delta}{T_s} > \frac{\Delta_{opt}}{T_s}$$

$$\boxed{\Delta > \Delta_{opt}}$$

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## Adaptive Delta modulation

- \* It is modified version of Delta modulation.
- \* To overcome Slope distortion & Granular in Delta modulation Adaptive Delta modulations used.
- \* Here the step size  $\Delta$  of the Quantizer is constant but varies with time.

Question (19) (b) Explain MSK Punctual.

Answer MSK - Minimum Shift Keying.

MSK sig can be represented as -

$$S(t) = S_1 \phi_1(t) + S_2 \phi_2(t)$$

→ Where coefficient of  $S_1$  and  $S_2$  are related to phase state  $\Theta(0)$  and  $(\Theta T_b)$  respectively -

$$* \phi_1(t) = \sqrt{\frac{2}{T_b}} \cos\left(\frac{\pi}{2T_b} t\right) \cos(2\pi f_c t). \quad T_b \leq t \leq T_b$$

$$* \phi_2(t) = \sqrt{\frac{2}{T_b}} \sin\left(\frac{\pi}{2T_b} t\right) \sin(2\pi f_c t) \quad 0 \leq t \leq 2T_b$$

→ To evaluate  $S_1$ , we integrate  $S(t) \phi_1(t)$  b/w bounds ( $-T_b$  &  $T_b$ )

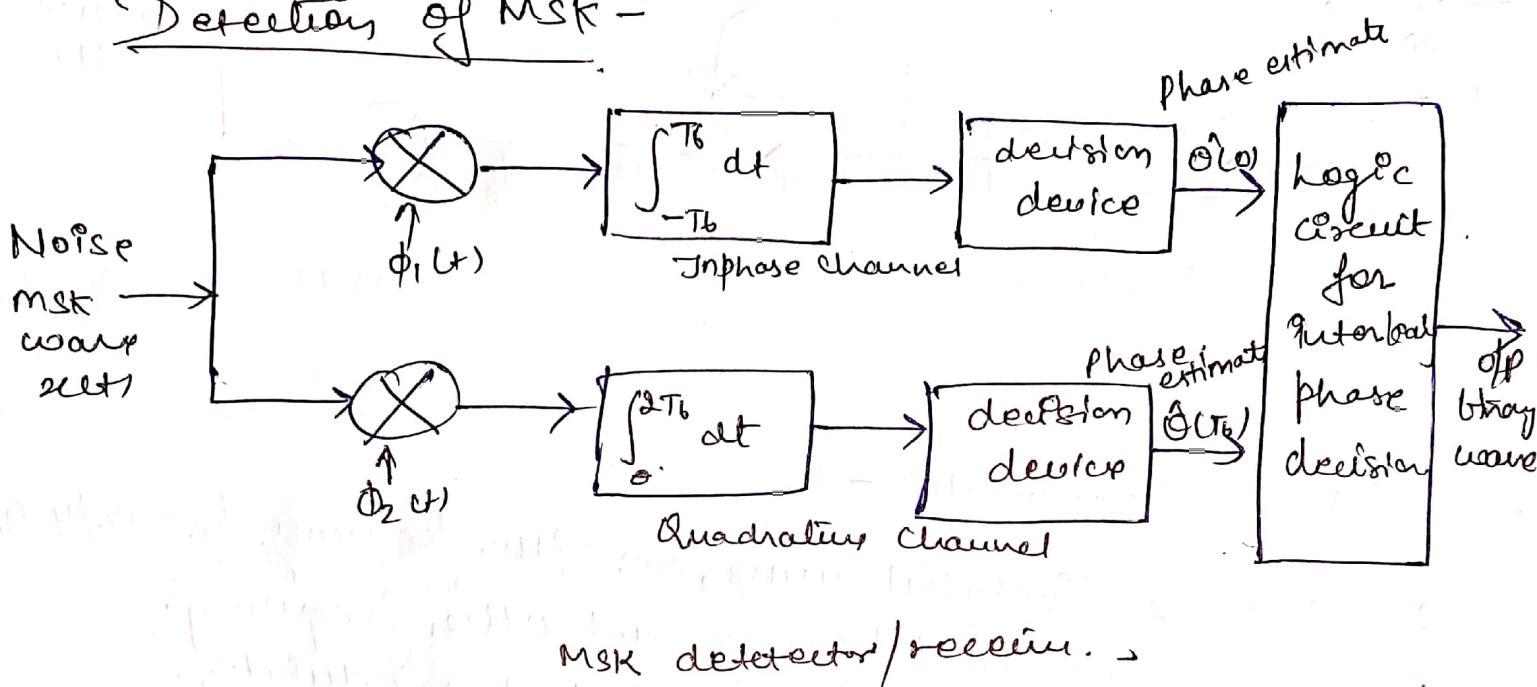
$$S_1 = \int_{-T_b}^{T_b} S(t) \phi_1(t) dt = \sqrt{E_b} \cos[\Theta(0)]$$

$$S_2 = \int_0^{2T_b} S(t) \phi_2(t) dt = -\sqrt{E_b} \sin[\Theta(T_b)]$$

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- finally  $\phi_1(t)$  and  $\phi_2(t)$  are multiplied with two binary wave  $m_1(t)$  and  $m_2(t)$ , both of which have bit rate equal to  $1/T_b$ .
- There two binary waves are extracted from incoming binary wave  $b(t)$ . At the o/p of summer, we have msk wave  $(s(t))$ .

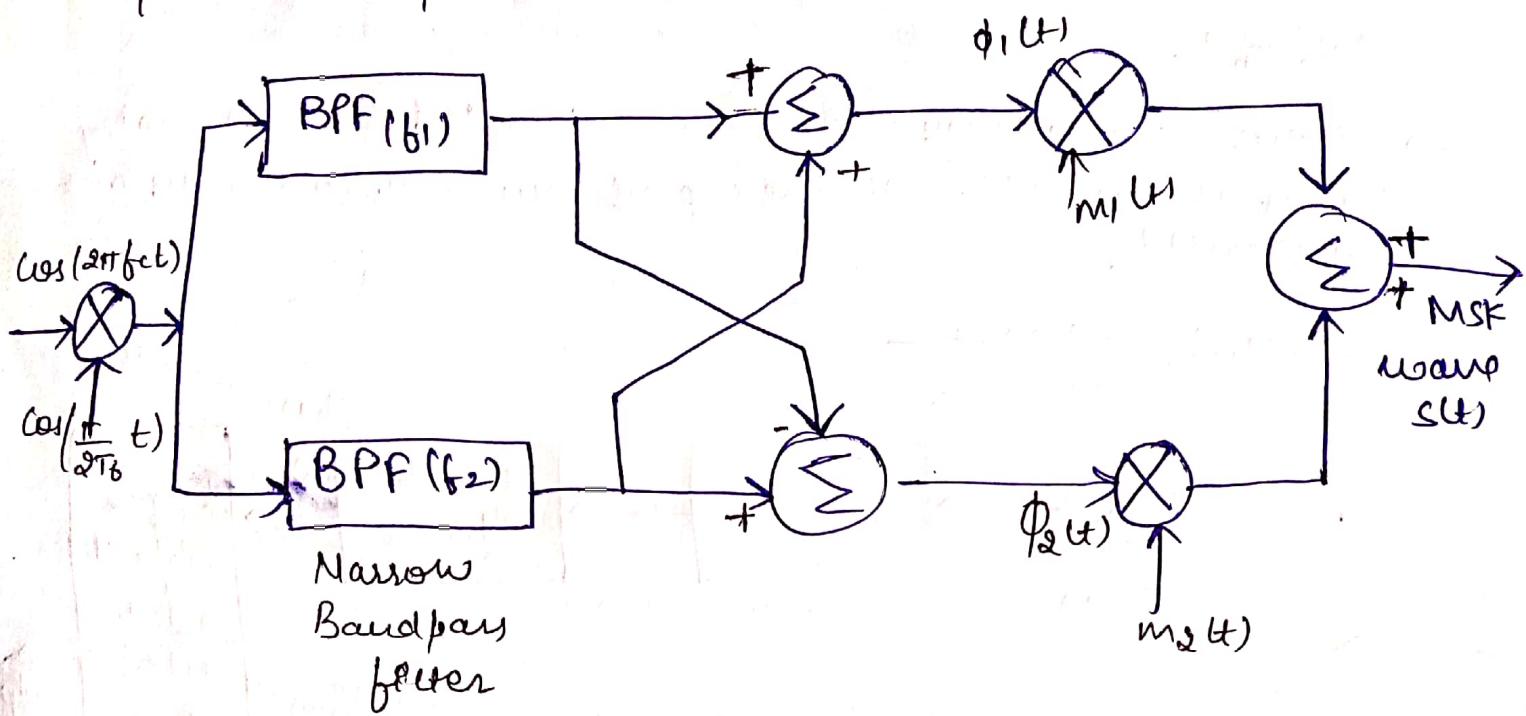
### Detection of MSK -



- The received Sig  $x(t)$  is correlated with locally generated of the coherent reference sig  $\phi_1(t)$  and  $\phi_2(t)$
- The resulting two channels o/p  $x_1$  and  $x_2$  are next compared w/ the threshold of zero volt & estimates of phase  $\phi_1(t)$  and  $\phi_2(t)$  are derived.
- finally these phase decision & Interleaved so as to construct original P/P binary wave  $b(t)$ .

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## Generation of MSK -



## MSK Transmitter -

- \* Two Q/P sinusoidal waves, one of the frequency  $f_c = n_c/4T_b$  for some fixed integer  $n_c$  and other frequency  $\frac{1}{4}T_b$  are first applied to product modulators.
- \* This produces two phase coherent sine wave at frequencies  $f_1$  and  $f_2$ , which are related to  $f_c$ , and bit rate  $1/T_b$ . 
$$f_c = \frac{1}{2} (f_1 + f_2)$$

deviation ratio

$$\delta = \frac{1}{2} T_b (f_1 - f_2)$$

- There two sinusoidal waves are separated from each other by two narrow band filter, one centered at  $f_1$  and other at  $f_2$ .
- Resulting filter o/p are next summed to produce the pair of quadrature carrier  $\phi_1(t)$  and  $\phi_2(t)$ .

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Question (17) (b) Explain the Quantization Error in detail & derive the expression for max. S/N to Noise Ratio in PCM system.

Answer Quantization error -

In uniform quantization, original continuous time function is first rounded off to several possible levels, yielding a staircase type of waveform. This introduces an error known as quantization noise/error. i.e. the staircase approximation is not identical to the original continuous sig. It is the difference b/w the two represents an error, known as quantization noise.

$$\rightarrow \text{For a midrise, } q_e = -\frac{\Delta}{2} \leq q_e \leq \frac{\Delta}{2}$$

Assuming that error is equally likely to lie anywhere in range  $(-\frac{\Delta}{2}, \frac{\Delta}{2})$ , the mean square  $(q_e)^2$  is given as.

$$(q_e^2) = \frac{1}{\Delta} \int_{-\Delta/2}^{\Delta/2} q_e^2 \cdot dq_e = \frac{1}{\Delta} \left| \frac{q_e^3}{3} \right|_{-\Delta/2}^{\Delta/2} = \frac{1}{3\Delta} \left[ \frac{\Delta^3}{8} + \frac{\Delta^3}{8} \right] = \frac{\Delta^2}{12}$$

$$q_e^2 = \frac{\Delta^2}{12}$$

Now let peak to peak voltage is  $(mp \text{ to } mp)$ , be total no. of quantizing level are  $L$  then step size  $\Delta$  is given by

$$\Delta = 2 \frac{mp}{L}$$

put this value of quantizing levels in  $\langle q_e^2 \rangle$

$$\langle q_e^2 \rangle = \frac{4mp^2}{L^2 \times 12} \Rightarrow q_e^2 = \frac{mp^2}{3L^2}$$

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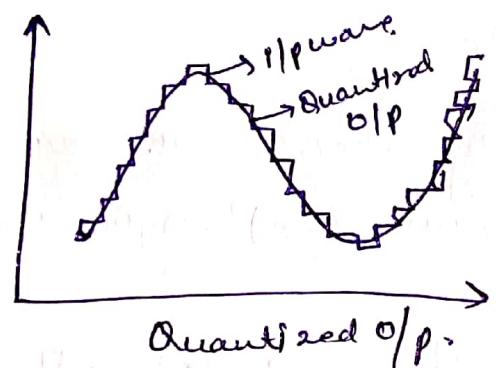
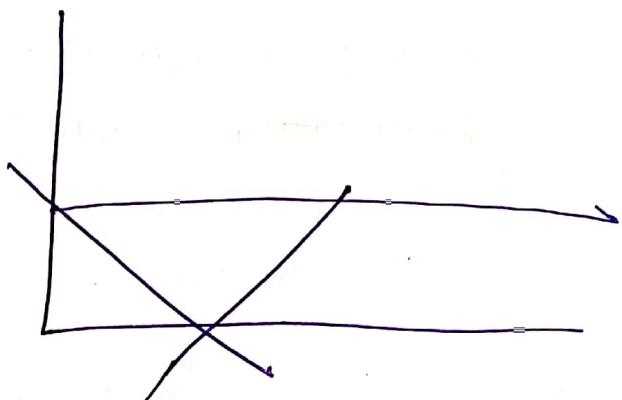
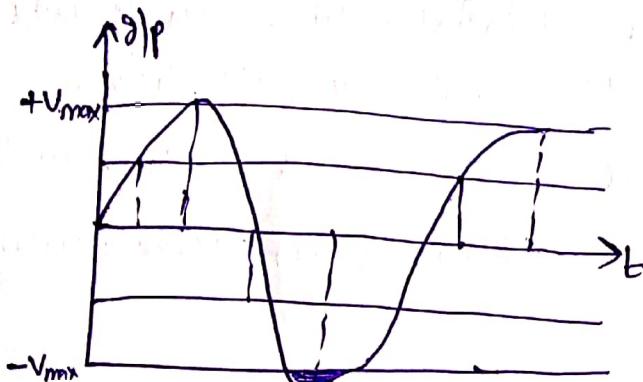
If the encoder used for decimal to binary conversion have  $n$  bits then total bits required to represent  $L$  levels can be given by

$$L = 2^n$$

correspondingly,  $n = \log_2 L$

Quantization error

$$(Qe^2) = \frac{m^2 p}{3 \cdot 2^{2n}}$$



## # Signal to Quantize Noise Ratio

$$\text{Sig to Quan. Noise Ratio} = \frac{\text{Normalised S/g Power}}{\text{Normalised quantize noise power}}$$

S/g power  $\Rightarrow$  for a single tone msg having peak value  $-mp_{\text{top}}$ , the power can be given by,

$$P = \frac{m^2 p^2}{2}$$

$$\left[ \because m(A) = mp \text{ (constant)} \right] \\ P = \frac{m^2 p^2}{2}$$

$$\text{Quantize Noise power} \Rightarrow Qe = \frac{m^2 p^2}{3 \cdot 2^n}$$

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$$SQNR = \frac{m_p^2}{\frac{2}{m_p^2}} = \frac{3L^2}{2} \Rightarrow SQNR = \frac{3L^2}{2}$$

$$SQNR \text{ in dB} \Rightarrow SQNR \text{ in dB} = 10 \log_{10} \frac{3L^2}{2}$$

$$\therefore SQNR \text{ in dB} = 20 \log_{10} \frac{3 \cdot 2^n}{2} = 1.76 + 6.02 n$$

$\left\{ SQNR \cong 1.8 + 6n \right\}.$

### Question 20 (b)

Ques Calculate the Signal to Noise ratio of FM. Also calculate figure of merit.

Answer-

Signal to Noise ratio -

It is defined as the ratio of sig power to noise power at the same pt.

Therefore,

$$\frac{S}{N} = \frac{X_s}{X_n} = \frac{V_s^2 / R}{V_n^2 / R} = \left( \frac{V_s}{V_n} \right)^2$$

where,

$S$  = Sig power.

$N$  = Noise power

→ The comparison of  $S/N$  ratio of I/p & O/p of a pose N/w problem, the noisiness duplication of the N/w.

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## Noise In FM System -

Let us consider that O/P sig to the IF carrier filter -

$$S_1(t) = A \cos [\omega_c(t) + K \int_{-\infty}^t m(\tau) d\tau]$$

$m(t)$  = frequency modulating baseband wavefor.

→ we assume that S/Ig is embedded in addition white gaussian noise of power-spectral density  $n/2$ .

→ IF carrier filter has a b-w  $B = 2\Delta f + 2f_m$

→ This filter passes the S/Ig with negligible distortion & eliminates all noise outside the b-w ( $B$ ).

→ When S-N Ratio is high ↑, the noise does not affect the O/P signal power.

→ O/P sig arrives at O/P -

$$S_2(t) = A_L \cos [\omega_c(t) + K \int_{-\infty}^t m(\tau) d\tau]$$

$$\phi(t) = K \int_{-\infty}^t m(\tau) d\tau$$

→ For O/P discriminator -

$$S_3(t) = \alpha \omega_c + \alpha K m(t)$$

$$\text{O/P sig} \rightarrow S_3(t) = \alpha K m(t)$$

$$\text{O/P sig power} = S_3(t) = \alpha^2 K^2 m^2(t)$$

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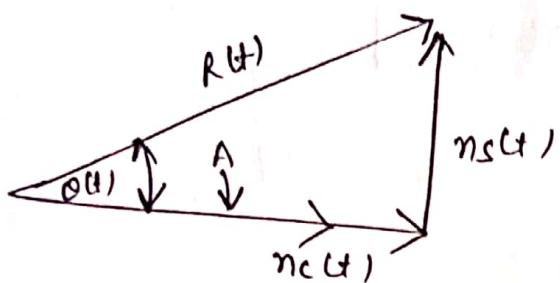
Output Noise power -

→  $\text{J/p}$  white noise having a power spectral density  $n/2$ .

→  $n_e(t) = 0$ , i.e. noise  $\sigma_p \cong$  independent of  $w_c(t)$

→ At Limiter  $\text{J/p}$ .

$$\begin{aligned} V_p(t) &= A \cos w_c(t) + n_c(t) \cos w_c(t) - n_s(t) \sin w_c(t) \\ &= [A + n_c(t)] \cos w_c t - n_s(t) \sin w_c t \end{aligned}$$



phasor represents  $n_s(t) \cos w_c t$ . In phase with the carrier phasor  $A \cos w_c t$ .

$$R(t) = \sqrt{(A + n_c(t))^2 + n_s(t)^2}$$

Similarly, phase,  $\Theta(t) = \tan^{-1} \frac{n_s(t)}{A + n_c(t)}$

Signal & noise forming  $V_1(t) = R(t) \cos [w_c t + \Theta(t)]$

$$V_1(t) = A_L \cos [w_c t + \Theta(t)]$$

Let us assume -

$$\Theta(t) = \frac{n_s(t)}{A}, \text{ Then } V_1(t) = A_L \cos \left[ w_c t + \frac{n_s(t)}{A} \right]$$

Compare eqn -

$$V_1(t) = \alpha w_c t + \alpha \frac{d}{dt} \phi(t)$$

$$V_1(t) = \alpha \left[ w_c t + \frac{1}{A} \frac{d}{dt} n_s(t) \right]$$

Avoid DC term -

$$V_1(t) = \frac{\alpha}{A} \frac{d}{dt} n_s(t).$$

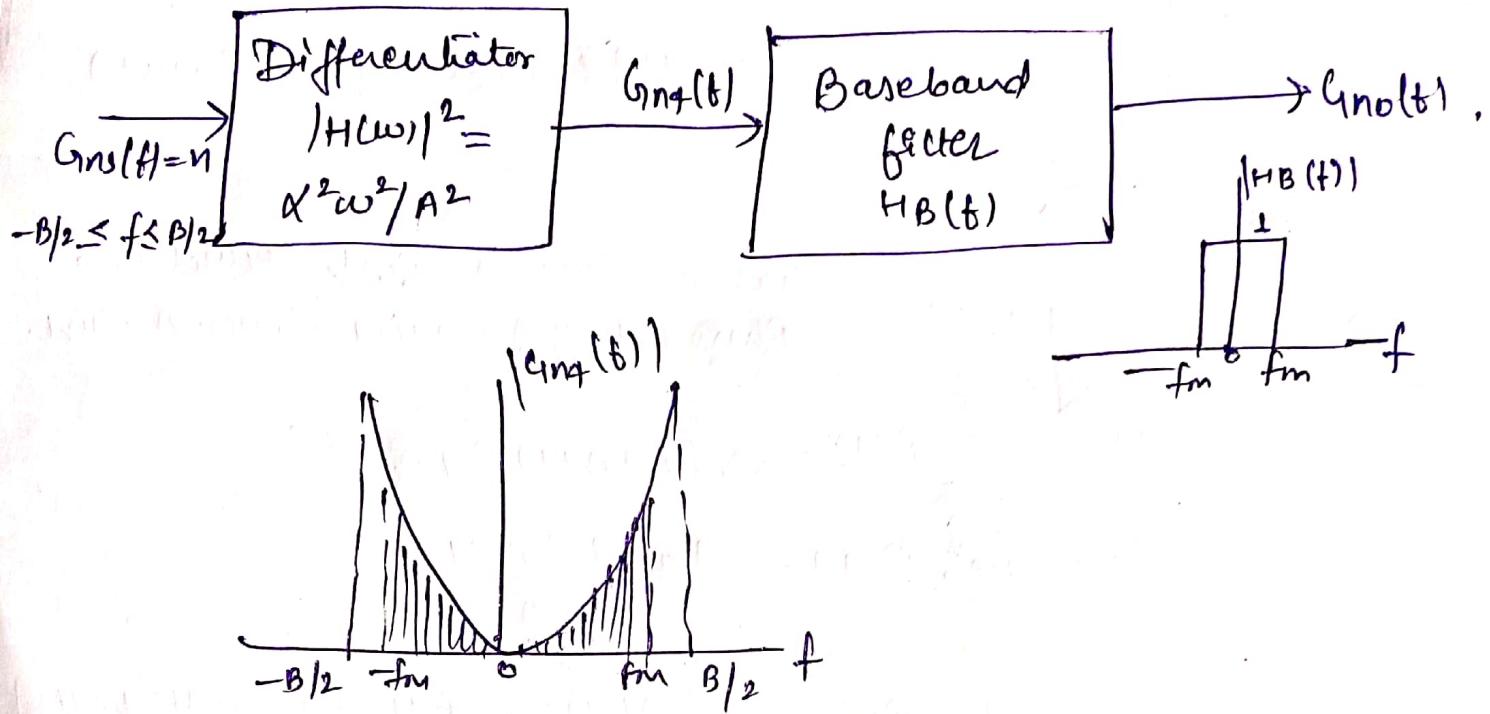
$n_s(t)$  Ps  $G_{n_s}(f)$  given by -

$$G_{n_s}(f) = \frac{\alpha^2 \omega^2}{A^2} n, |f| < \frac{B}{2}$$

Y-axis

O/p noise power (No)

$$N_o = \int_{-B/2}^{B/2} G_{nq}(f) \cdot df = \frac{\alpha^2 n}{A^2} \int_{fm}^{fm} 4\pi^2 f^2 df \\ = \frac{8\pi^2}{3} \cdot \frac{\alpha^2 n}{A^2} f^3 m.$$

O/p SNR -

The o/p SNR can be computed as -

$$\frac{S_o}{N_o} = \frac{\alpha^2 K^2 \overline{m^2(t)}}{\left(\frac{8\pi^2}{3}\right) \left(\frac{\alpha^2 n}{A^2}\right) f^3 m} = \frac{3 K^2 \overline{m^2(t)} \cdot f^2 / 2}{4\pi^2 \cdot f_m^2 \cdot n f_m}$$

$m(t)$  ps sinusoid & produce a  $\Delta f$ .

$$S(t) = A \cos \left[ \omega_c(t) + \frac{\Delta f}{f_m} \sin 2\pi f_m t \right] \xrightarrow{\text{modulating frequency.}}$$

Comparing eqn

$$S(t) = A \cos \left[ \omega_c(t) + R \int_{-\infty}^t m(\tau) d\tau \right], \text{ with eq above}$$

$$\text{i.e. } Rm(t) = 2\pi \Delta f \cos 2\pi f_m t$$

$$R^2 \overline{m^2(t)} = \frac{4\pi^2 (\Delta f)^2}{2} = 2\pi^2 (\Delta f)^2$$

*Karthy*

Putting value of  $R^2 m^2 u^2 \ln 2$  in eq.

$$\frac{S_o}{N_o} = \frac{3}{2} \left( \frac{\Delta f}{f_m} \right)^2 \cdot \frac{n^2/2}{n f_m} = \frac{3}{2} B^2 \frac{S_i}{N_m}$$

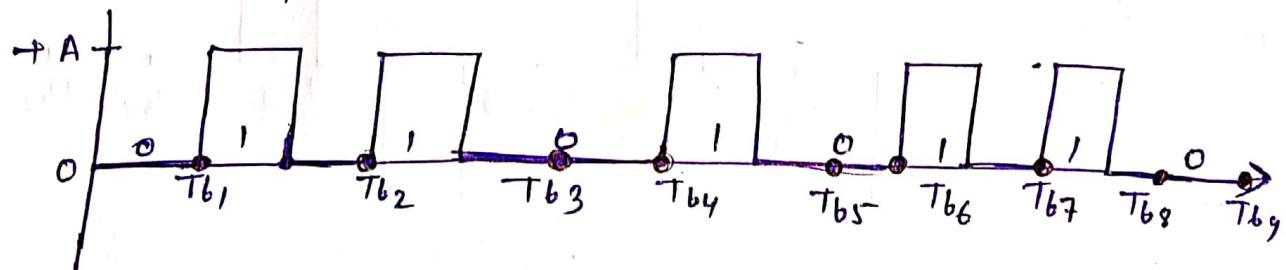
Here,  $\boxed{\beta = \frac{\Delta f}{f_m} = \text{modulation Index}}$

$$Y_{fm} = \frac{\frac{S_o}{N_o}}{\frac{S_i}{N_m}} = \boxed{\frac{3}{2} \beta^2}$$

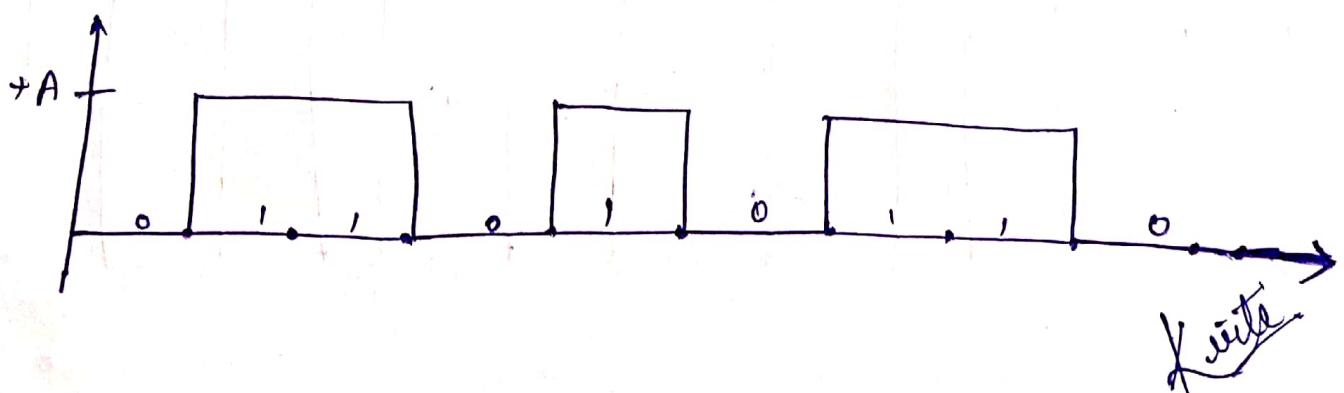
Question - ~~Q18~~ 18 (a)

Quesn. for a given binary sequence. 011010110, construct Unipolar RZ, Unipolar NRZ, polar RZ, Polar NRZ, AMI & Manchester format.

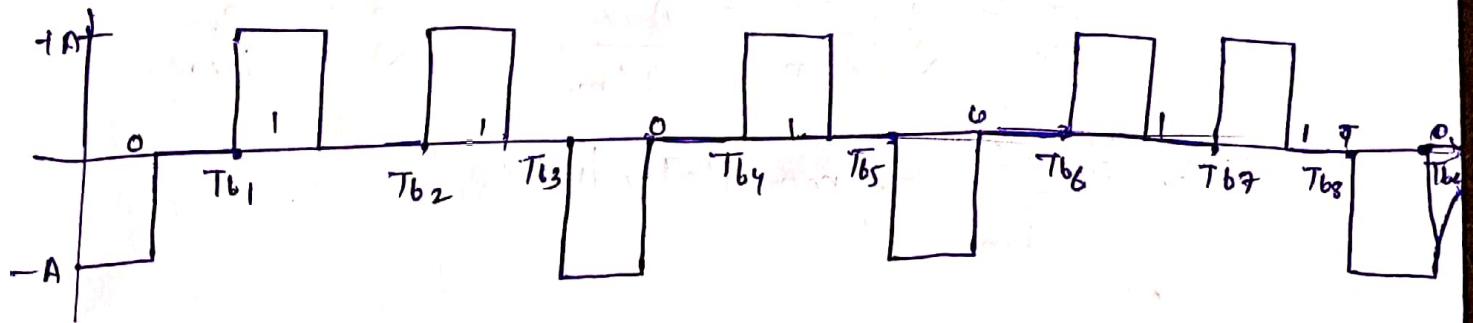
Answer - (i) Unipolar RZ



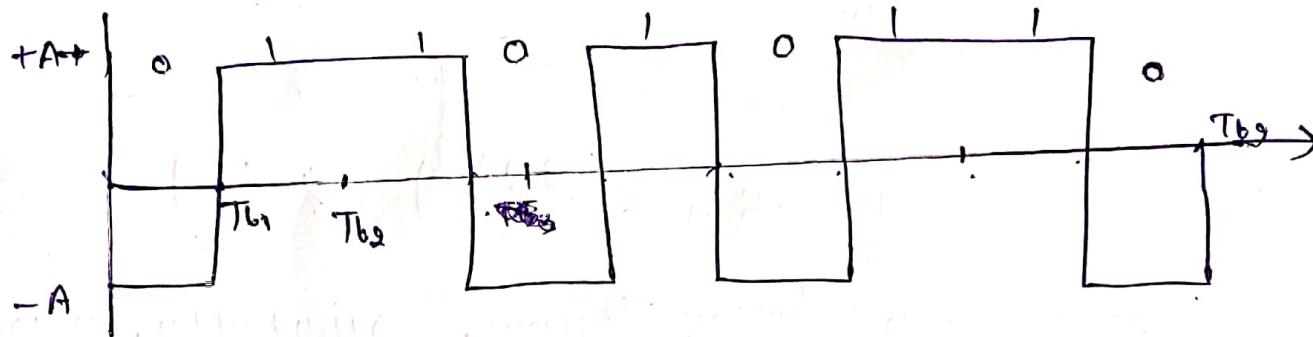
(ii) Unipolar NRZ



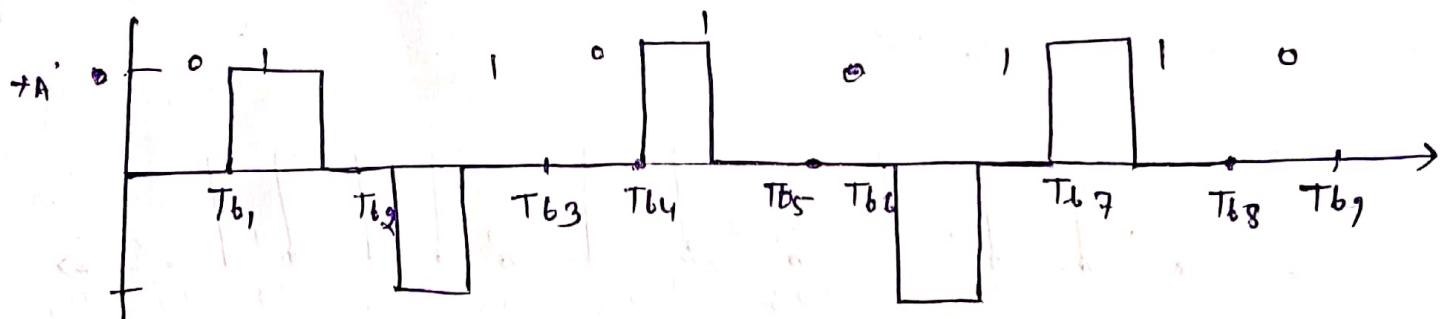
(iii) Polar RZ



(iv) polar NRZ



(v) AM



(vi) Manchester

