

Digital communication

Continuous is also called Analog Signal, where demodulation is difficult, distortion is happening.

Converting Analog to digital and these we will transmit we have regeneration block.

- * Noise can't distort digital signal and digital is immune to noise.

- * Analog is used in 1G communication.

- * 2G - Till now [digital]

- * Threshold value is fixed greater than it's 1 less than it is zero.

Sampling

It is a process which we convert analog (continuous) signal to digital signal.

Analog



Sampling



Quantization



Encoding



Digital

- * Sampling Theorem states that Sampling frequency must be greater than twice f_m

$$f_s > 2f_m \rightarrow \text{Nyquist Rate}$$

f_m = Maximum frequency of message signal.

- * Major criteria for sampling.

$$m(t) = 1 + \sin 100\pi t + \cos 200\pi t + \sin 150\pi t$$

$\omega = 2\pi f$ find frequency and highest frequency is considered as fm.

$$\boxed{f_s > 2fm}$$

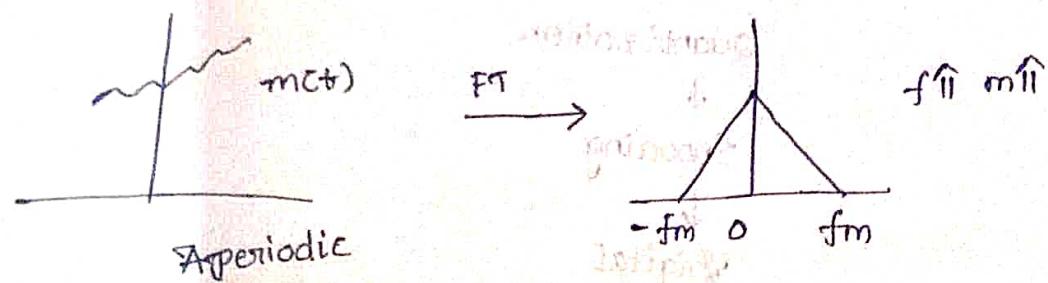
$$\boxed{T_s < \frac{1}{2fm}}$$

$$f_s > 200\text{Hz}$$

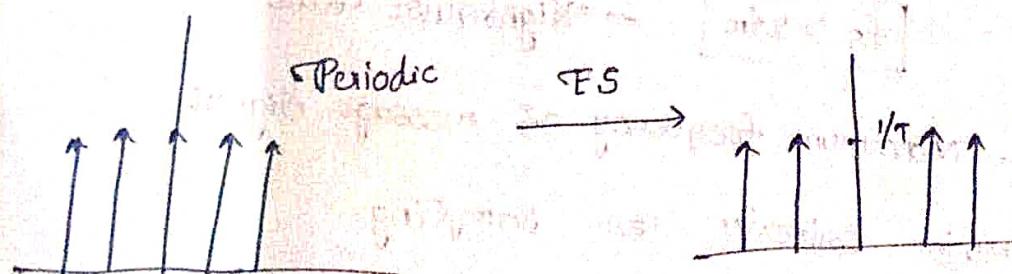
Sampling →

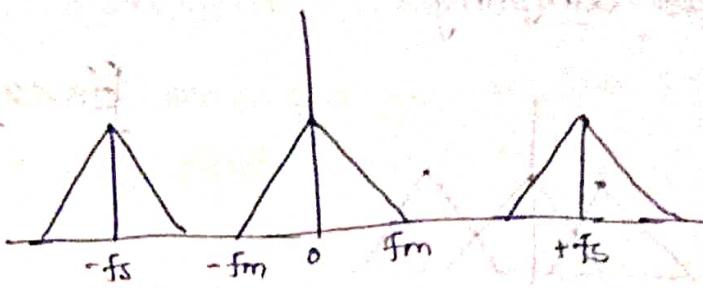
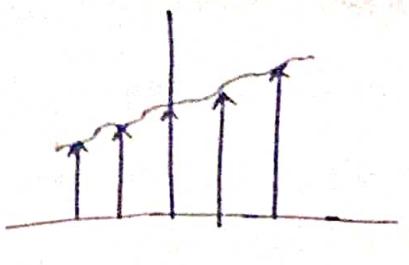
- over Sampling $f_s > 2fm$
- Critical Sampling $f_s = 2fm$
- under Sampling $f_s < 2fm$

Over Sampling



multiply it with an impulse then we get





$\text{anything} \times \text{Impulse} = \text{Anything}$

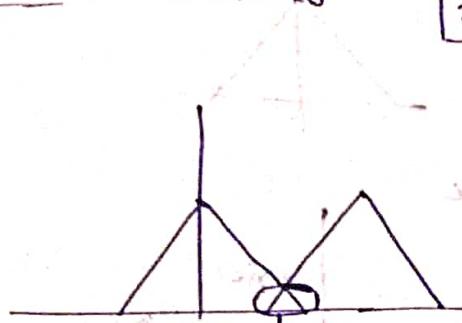
Shifted version of $\times \text{Impulse} = \text{shifted version of anything}$

$f_s > 2f_m$

We can easily get back the original signal.

Critical Sampling

$f_s = 2f_m$

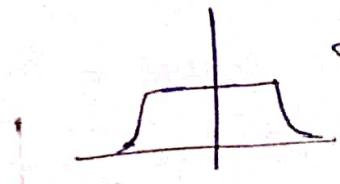
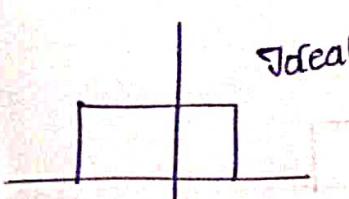


Some part is overlapping. Some sort of distortion

In case of critical sampling

is happening.

we can get two types of

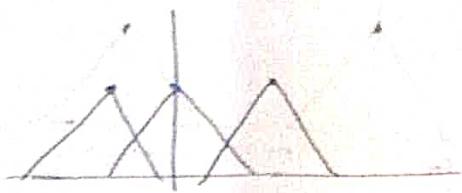


Practical

where practically it is not possible and theoretically possible.

Theoretically possible.

Under Sampling



$$f_s < 2f_m$$

overlapping of msg signal is called as

Aliasing Filter

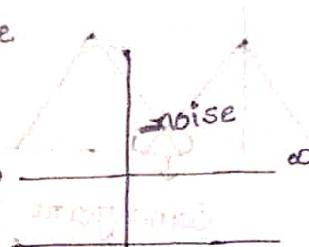
- * we can't bring back the original signal
- * So we always prefer over Sampling.

Anti-Aliasing Filter

* In order to limit the signal we will pass it through a LPF.



* But we can't avoid noise inside the range.

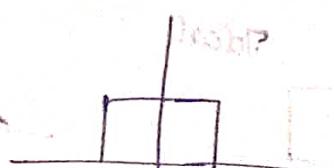


* That LPF is also known as Anti-Aliasing Filter

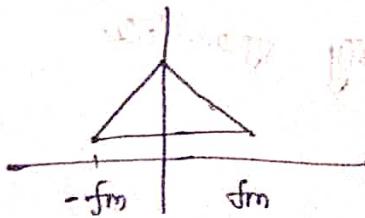


Q

Digital Filter

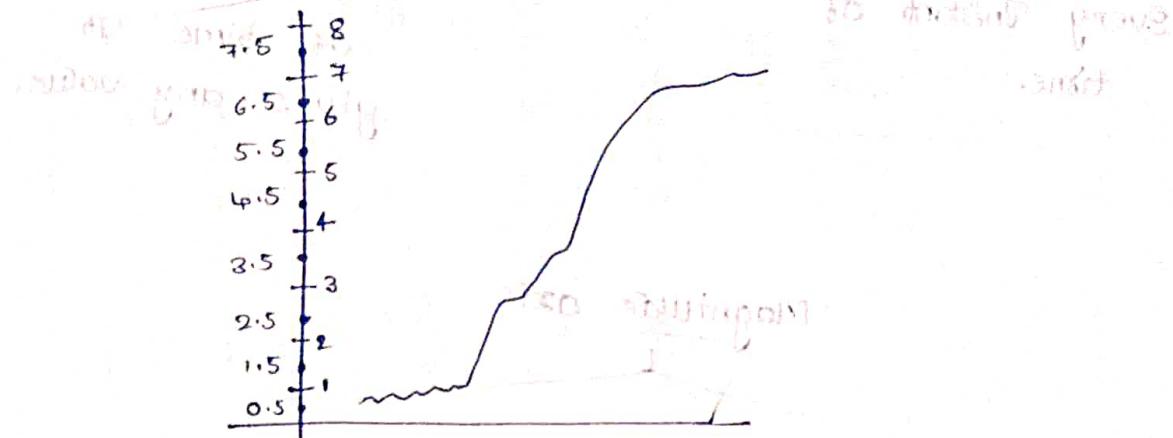


How do we get rid of the noise?



Quantization [utilise less memory] To save the memory.

It is a process by which we can divide entire range to segments and its mid points are assigned with codes [000, 001, ..., ...].



If we increase the length of the segments then accuracy is also increasing. For this case we require more memory & more time. we have to store each and every element with a particular code.

Encoding we encode the signal, we assign the quantized values with a particular code.

Quantization Error = Actual value - Quantization value.

* We can't completely remove the error But we can reduce the error by increasing accuracy and decrease the range of signal.

$$\text{Maximum Quantization Error} = \pm \frac{\Delta}{2}$$

Δ = segment length.

$$[8] = 2^3 [000, 001, 010, \dots]$$

Time axis

Every instant of time.

Discrete

Particular instance of time It gives any value.

Magnitude axis

Analog
Magnitude could be anything within the range.

Digital

Some particular value, specific set of values.

$$m(t) = \sin 200\pi t * \cos 400\pi t$$

$$\sin A * \cos B$$

$$\sin(A+B) + \sin(A-B)$$

$$= \sin 600\pi t + \sin 200\pi t$$

[Convert two frequencies into single term and then we have to find the maximum.]

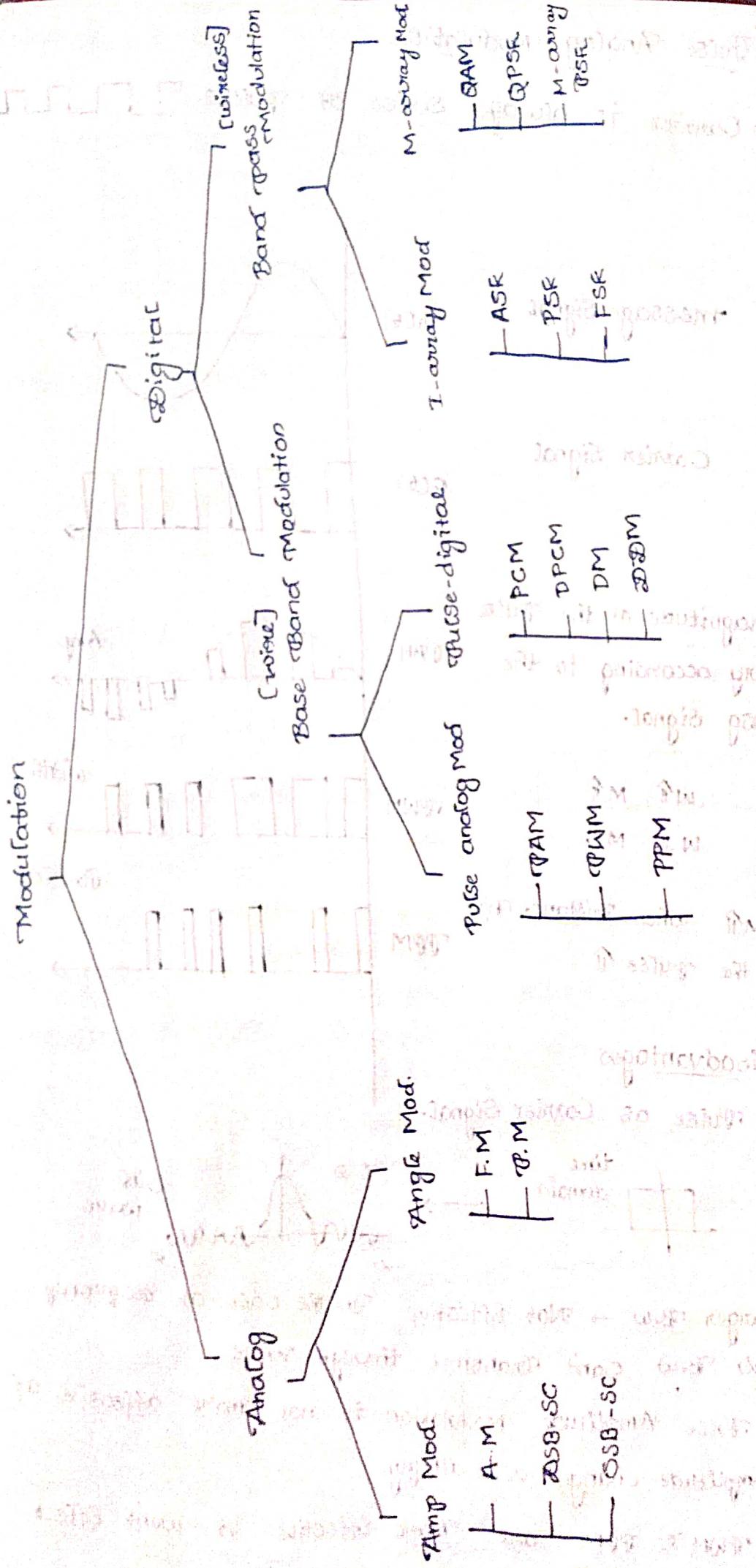
Substitute the term, divide and then do it again.

$$2\pi f = 600\pi$$

$$f_m = 300 \text{ Hz}$$

$$f_s > 2 \cdot f_m$$

$$f_s > 600 \text{ Hz}$$



Pulse Analog Modulation

* Carrier is always series of pulses.

message Signal

Carrier Signal

magnitude of the pulse vary according to the msg signal.

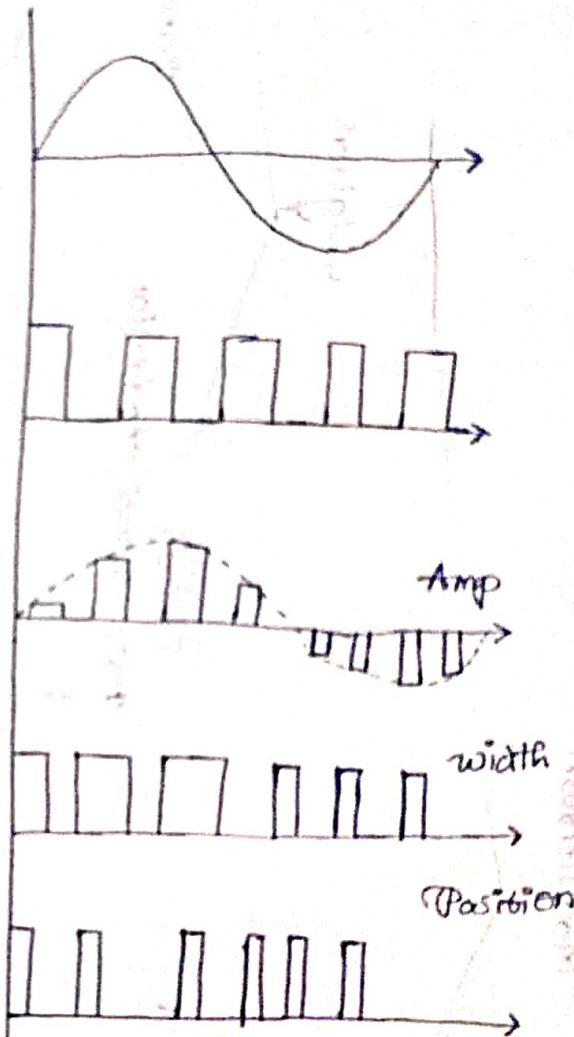
W ↑ M ↑

W ↓ M ↓

M ↑ when distance b/w the pulse ↑

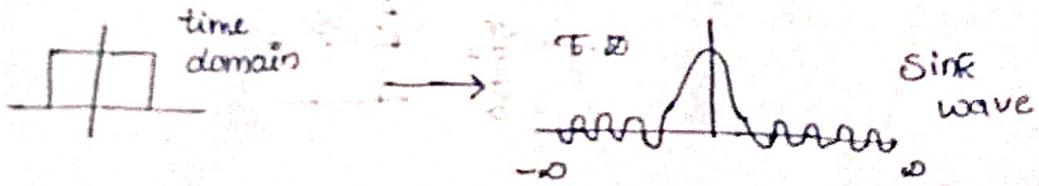
CPWM

PPM



Disadvantages

* Pulse as Carrier Signal.



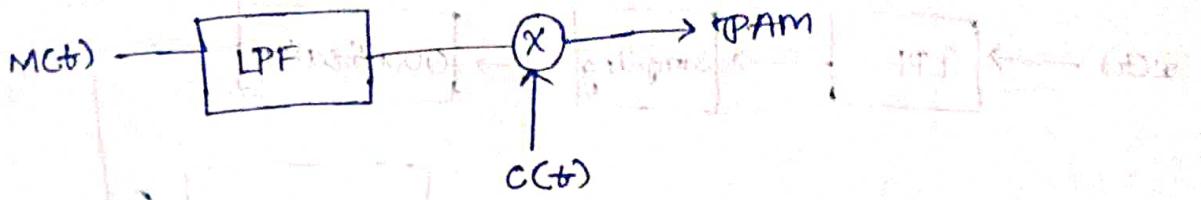
larger B.W \rightarrow Not Effective In the case of Frequency

* ω B.W can't transmit through wire..

* Pulse Amplitude Modulation is not more effective as Amplitude changes accordingly.

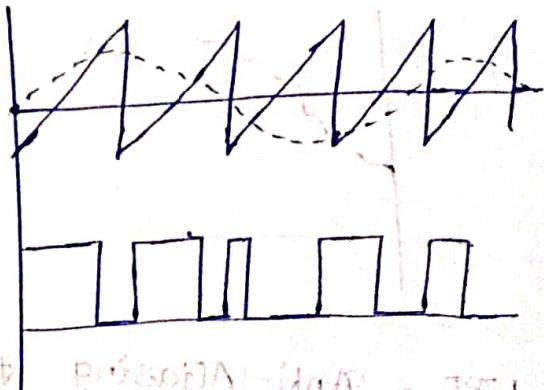
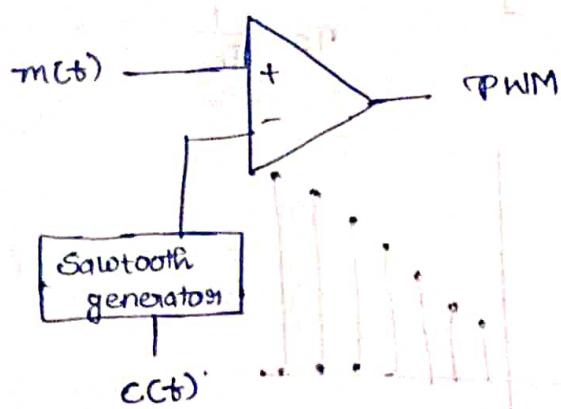
* PWM & PPM are More effective as it won't effect magnitude changes.

Modulation of PAM



LPF is used as Anti-Aliasing to avoid noise.

Modulation of PWM

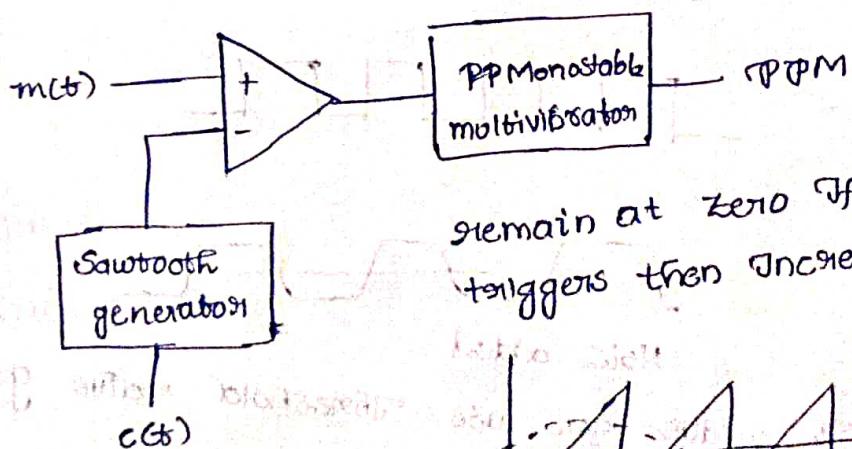


$m >$ sawtooth signal output = HIGH

$m <$ sawtooth signal output = LOW

Op-amp act as Comparator.

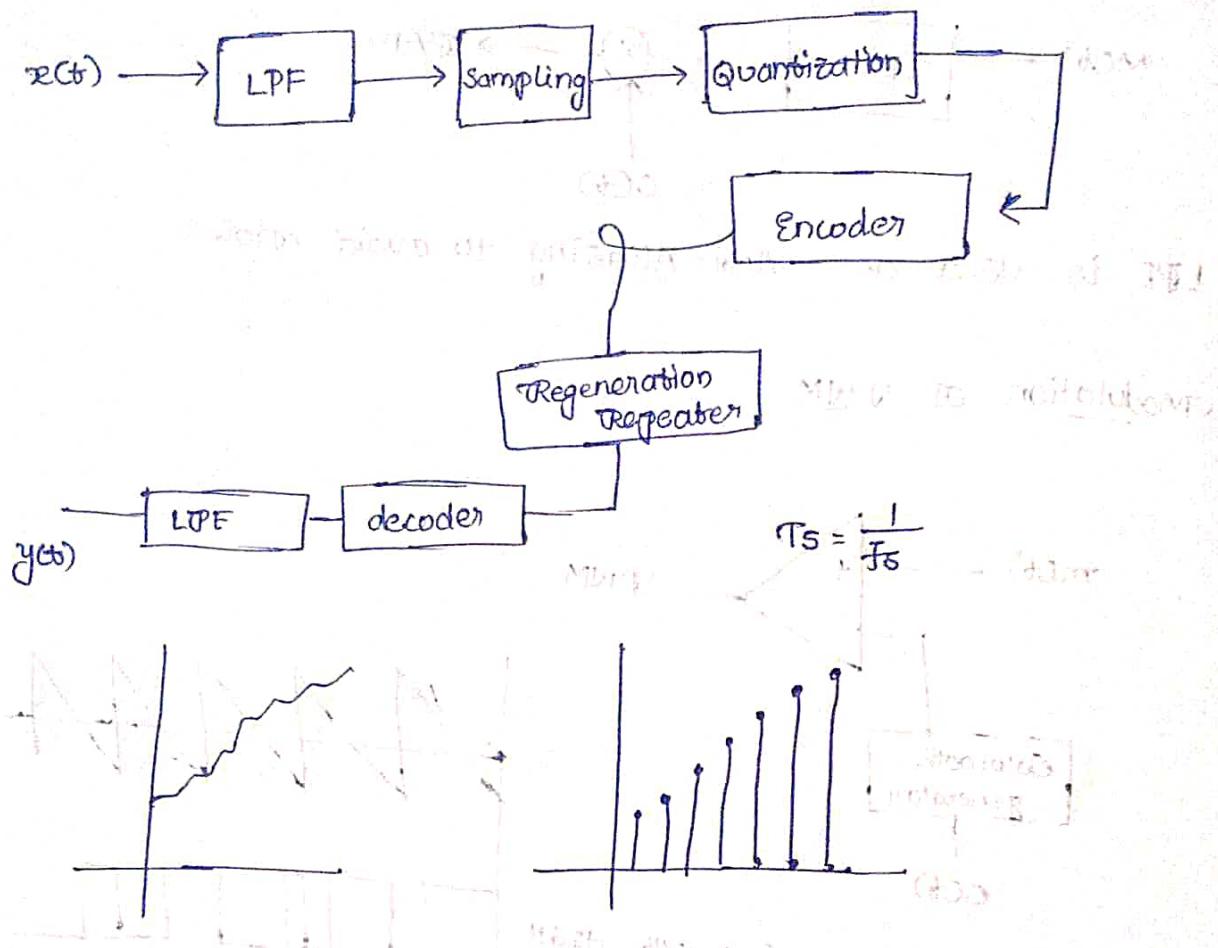
Modulation of PTPM



remain at zero if any input triggers then increase.



Pulse Code Modulation

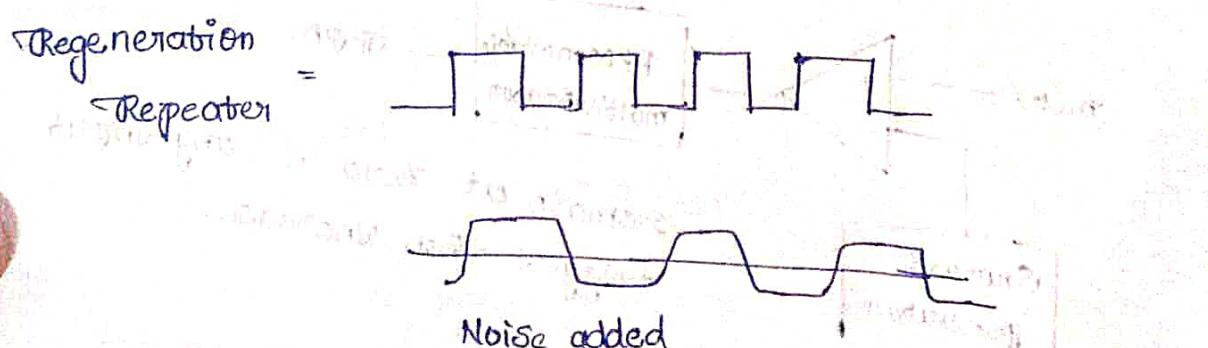


LPF = Anti-Aliasing Filter → to reduce B.W.

Sampling = continuous → discrete

Quantization = Save Memory.

Encoder = assigns a particular value.

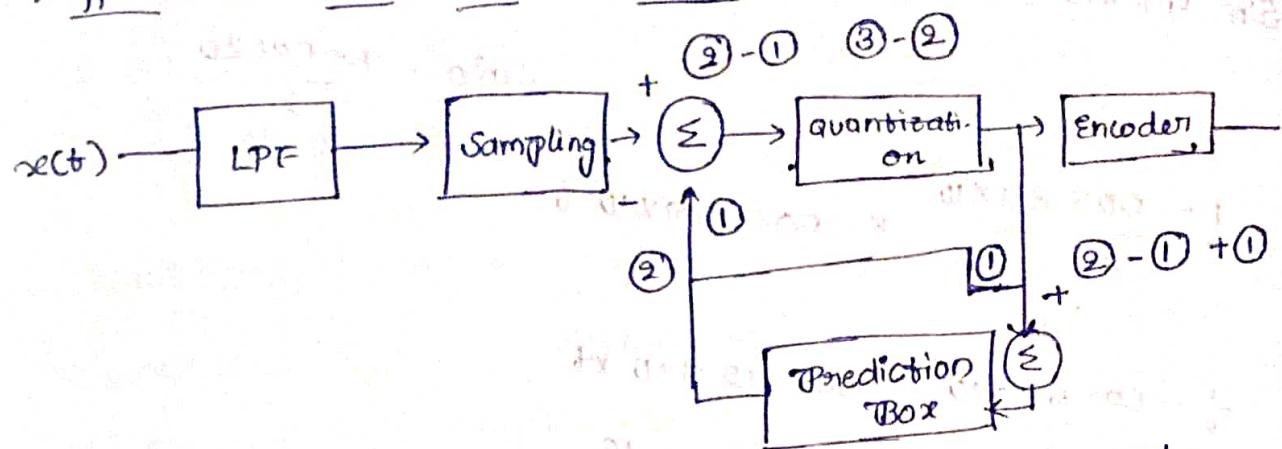


To reduce noise, use threshold value greater than it is less than, it is zero.

decoder → Decodes the signal

LPF → Again removes the noise added in Travelling.

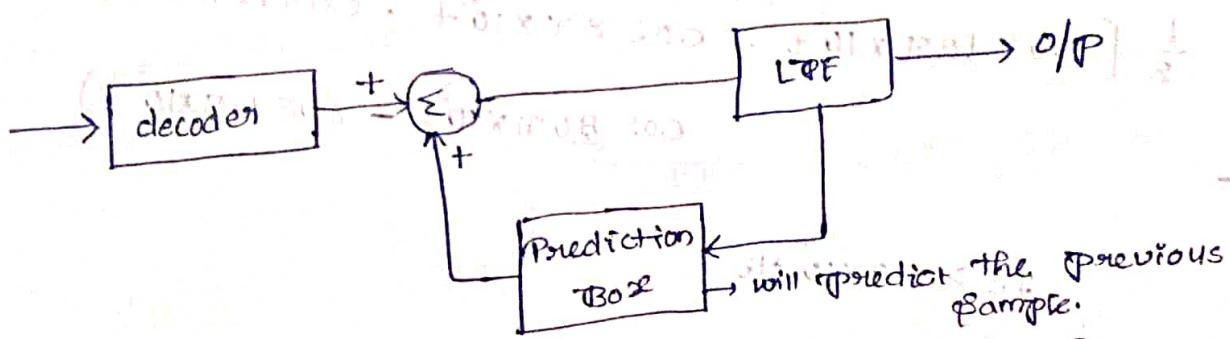
Differential Pulse Code Modulation



- * we don't transmit codeps we transmit diff RB/w two consecutive Signals [pointps]
- * where range \downarrow and quantization Error reduces.

SCI HUB

deModulation of DPCM



In Modulation we subtract the previous signal.
In deModulation we add the previous signal.

- $10 \sin 8\pi \times 10^3 t$ $\rightarrow f_s > 8 \times 10^3 \text{ Hz}$
- $6 \sin 4\pi \times 10^3 t + 8 \cos 12\pi \times 10^3 t \rightarrow f_s > 12 \times 10^3 \text{ Hz}$
- $\sin 4\pi \times 10^3 t \cdot \cos 12\pi \times 10^3 t \rightarrow f_s > 16 \times 10^3 \text{ Hz}$
- $\sin^2 4\pi \times 10^3 t \cdot \cos 12\pi \times 10^3 t \rightarrow$
- $\sin C 100t$
 $\sin 2\pi ft$

$$C = 3 \times 10^3$$

$$2\pi f = 8\pi \times 10^3$$

$$f$$

$$\sin^2 \theta = \frac{1 - \cos 2\theta}{2}$$

$$\sin(A+B) + \sin(A-B)$$

$$\sin^2 4\pi \times 10^3 t * \cos 12\pi \times 10^3 t$$

$$\sin^2 \theta = \frac{1 - \cos 2\theta}{2}$$

$$= \frac{1 - \cos 8\pi \times 10^3 t}{2} * \cos 12\pi \times 10^3 t$$

$$= \frac{1}{2} - \cos 4\pi \times 10^3 t \cdot \cos 12\pi \times 10^3 t$$

$$= \frac{1}{2} -$$

$$f_s > 16 \times 10^3 \text{ Hz}$$

$$= \cos \frac{A+B}{2} - \cos \frac{A-B}{2}$$

$$= \frac{1}{2} [\cos 12\pi \times 10^3 t - \cos 8\pi \times 10^3 t \cdot \cos 12\pi \times 10^3 t] \\ \quad - \cos 8\pi \times 10^3 t - \cos 4\pi \times 10^3 t)$$

=

$$f_s > 20 \times 10^3 \text{ Hz}$$

5) $\text{Sinc } 100t$ [Sinc width]

$$\boxed{\text{Sinc } 100t = \frac{\sin 100\pi t}{100\pi}}$$

$$2\pi f = 100\pi$$

$$f = 50 \text{ Hz}$$

$$\frac{1}{100\pi} \sin 100\pi t$$

$$f_s > 100 \text{ Hz}$$

n = no. of bits per Sample
 $L = 2^n$ [No. of Segments]

$$\Delta = \frac{V_{\max} - V_{\min}}{L}$$

$$\text{Max Quantisation Error} = \pm \frac{\Delta}{2}$$

Quantisation Error = Actual Value - Quantised Value.

$$\text{Bit duration} = \frac{\text{Sampling duration (Period)}}{\text{No. of Bits per Sample}}$$

$$f_s = \frac{1}{T_s}$$

$$\text{Bit rate} = \frac{1}{\text{Bit duration}}$$

(Speed of data per second)

Band width = Bit rate

1. A message Signal $m(t) = 10 \cos 8\pi \times 10^3 t$ is transmitted using pulse code modulation system, find all the parameters of PCM System.

$$n = 4$$

$$L = 16$$

$$\Delta = \frac{10 - (-10)}{16}$$

$$\Delta = \frac{20}{16}$$

$$\text{Max Quantisation Error} = \pm \frac{20}{2} = \pm \frac{20}{32}$$

$$\text{Quantisation Error} = \frac{1}{8000}$$

$$\text{Bit duration} = \frac{1}{8000}$$

$$= \frac{1}{32000}$$

$$\text{Bit rate} = 32000$$

$$\text{Band width} = 32000$$

A Sinusoidal msg signal $y = \sin(4\pi \times 10^3 t)$ transmitted through 8 level PCM system, with sampling [n=3].

Frequency has 5 times of Nyquist rate.

1. Find all the parameters.

2. Given. Sampled value $\rightarrow 3.2V, -2.8V, -0.1V,$

$1.5V, 3.9V$ Find corresponding Quantised & Encoded output.

$$n=3 \rightarrow 8 \text{ levels}$$

$$L=8$$

$$T_s = \frac{1}{f_s}$$

$$\Delta = \frac{8}{8} = 1$$

$$\Delta = \pm 1$$

$$2fm = 20K$$

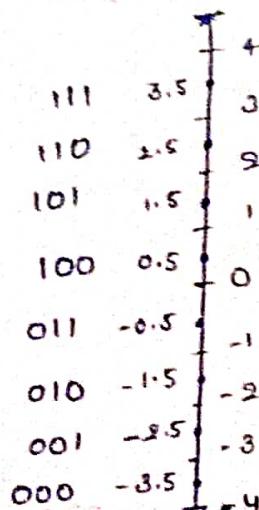
$$M.Q.E = \pm \frac{1}{2}$$

$$\frac{1}{20000}$$

$$\text{Bit duration} = \frac{1}{20000} = \frac{1}{60000}$$

$$\text{Bit rate} = 60K \text{ bits/sec} = \text{Band width.}$$

2. Dynamic range +4 to -4



-3.2	\rightarrow	-3.5	000
Actual Value		Quantized Value	$-3.2 + 3.5$
			Q.E = +0.3
-2.8	\rightarrow	-2.5	001
			Q.E = -0.3
-0.1	\rightarrow	-0.5	011
			$-0.1 + 0.5$
1.5	\rightarrow	1.5	101
			Q.E = 0
3.9	\rightarrow	3.5	111
			Q.E = 0.4

once quantised value w.e
can getolve.

$$\downarrow [Qc]_{\text{max}} = \pm \frac{\Delta}{2} \downarrow$$

$$\downarrow \Delta = \frac{V_{\text{max}} - V_{\text{min}}}{2^n} \downarrow$$

line Coding:

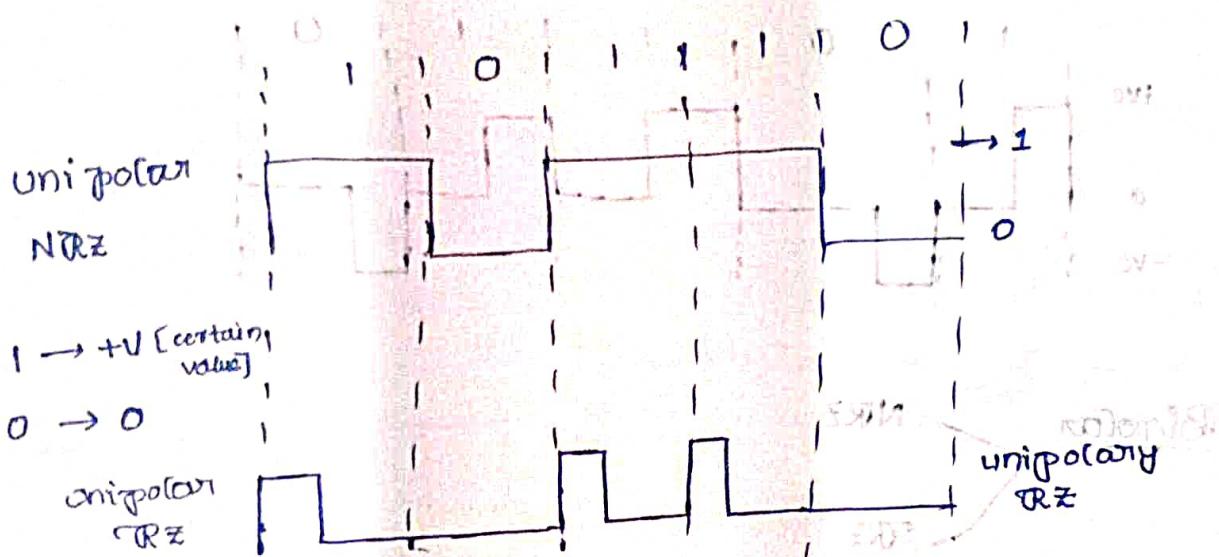
It is a way to represent bits in the form of signals.

1. unipolar (ON-OFF switching)

2. polar

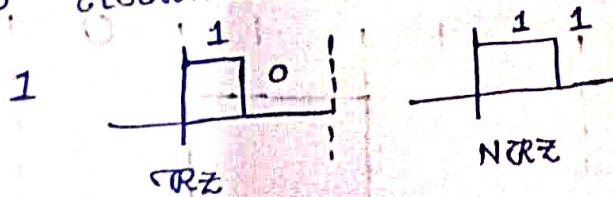
3. bipolar

Non-returning zero
returning zero



unipolar RZ

0 → ON
1 → +V for first $T_B/2$ duration T_B = Bit duration.
0 elsewhere



Polar

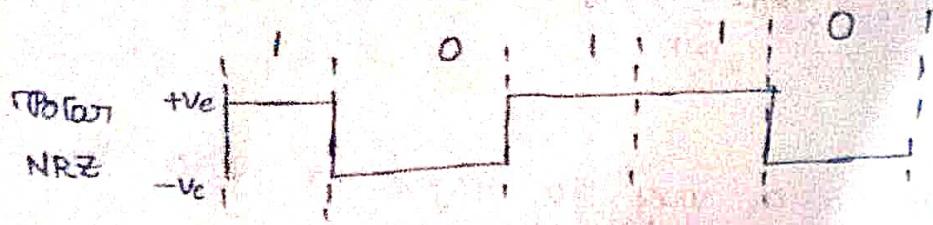
NRZ
RZ

$$NRZ - 0 = -V$$

$$1 = +V$$

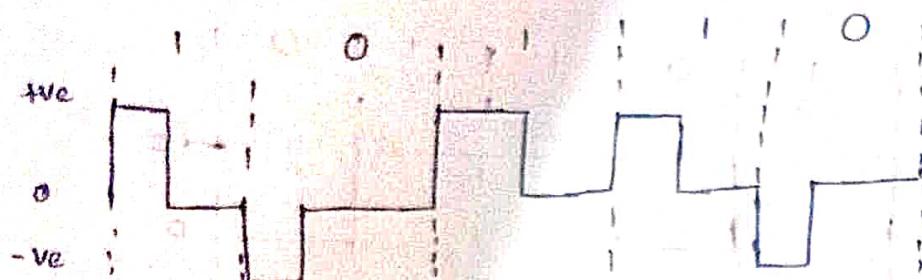
NRZ

$$\begin{cases} 1 & +v \\ 0 & -v \end{cases}$$



RZ

$$\begin{cases} 0 & -ve \text{ for } T_{6/2} \\ 0 & \text{otherwise} \\ 1 & +ve \text{ for } T_{6/2} \text{ duration} \\ 0 & \text{otherwise} \end{cases}$$



Bipolar

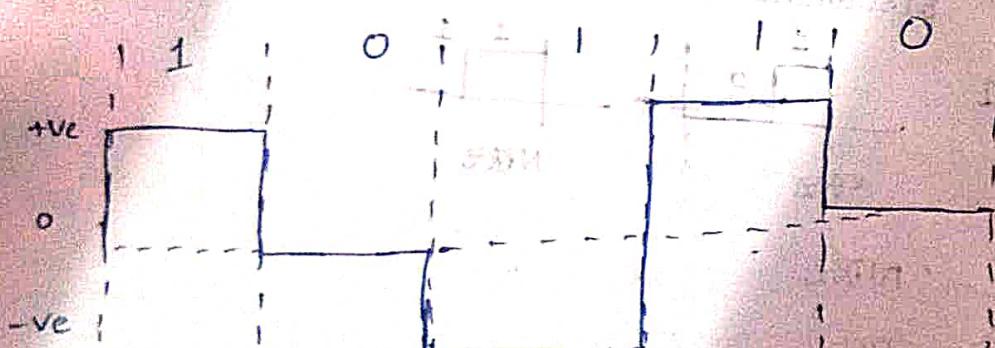
NRZ

RZ

$$0 \rightarrow OV$$

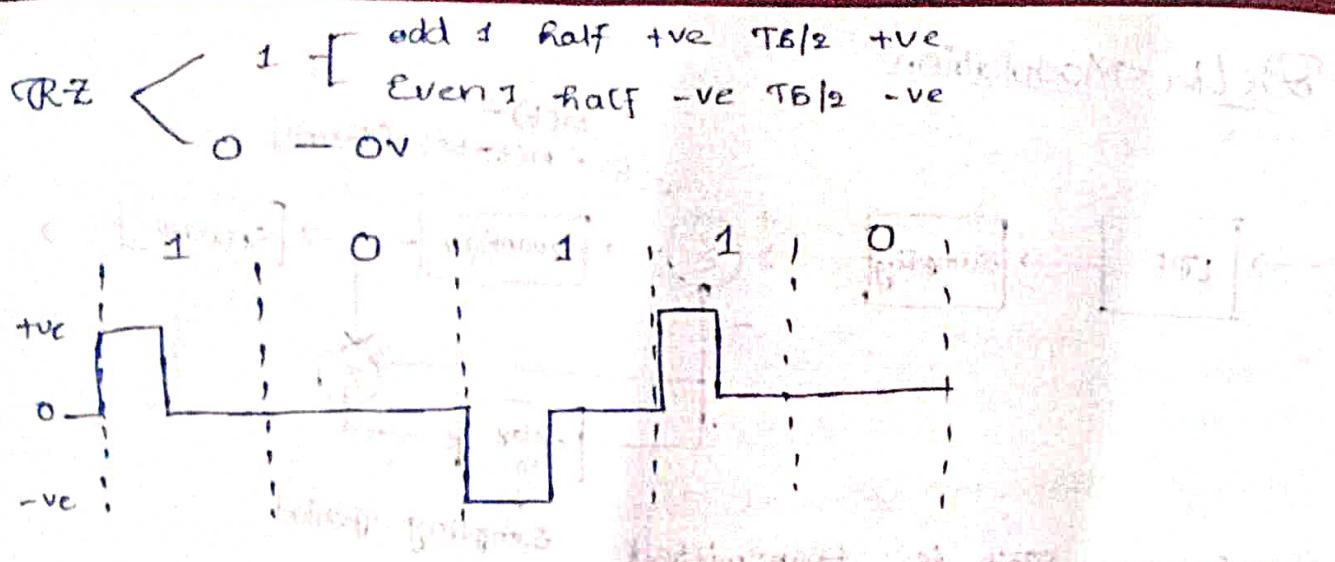
NTRZ

$$\begin{cases} 1 & \begin{cases} 1 & \text{positive} \\ 2 & \text{negative} \end{cases} \\ 0 & \text{odd } 1 \rightarrow +ve \\ 0 & \text{Even } 1 \rightarrow -ve \end{cases}$$

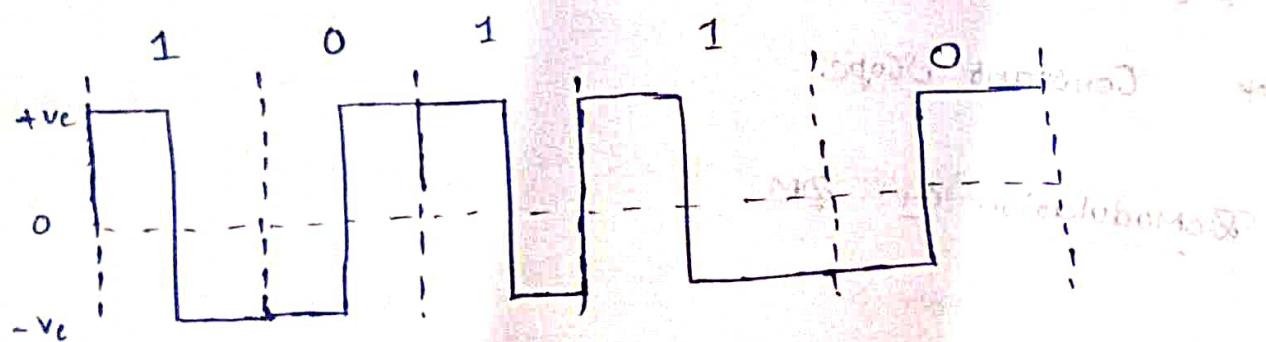
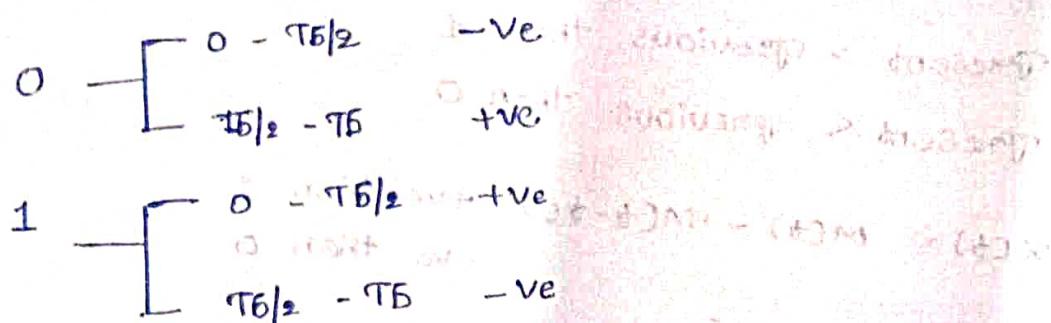


Alternatively

OKabi positive +v Inforati -v Ki veltadi



Manchester Coding



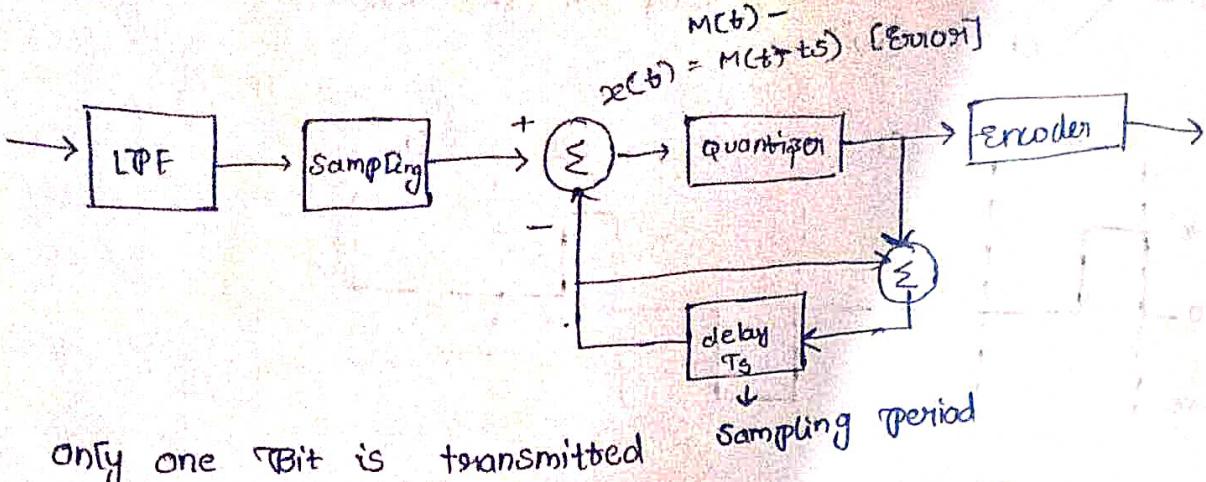
1. Channel noise → Regenerator, Repeater

2. Quantised noise

- code length ↑
- range ↓



Delta Modulation



present > previous then 1

present < previous then 0

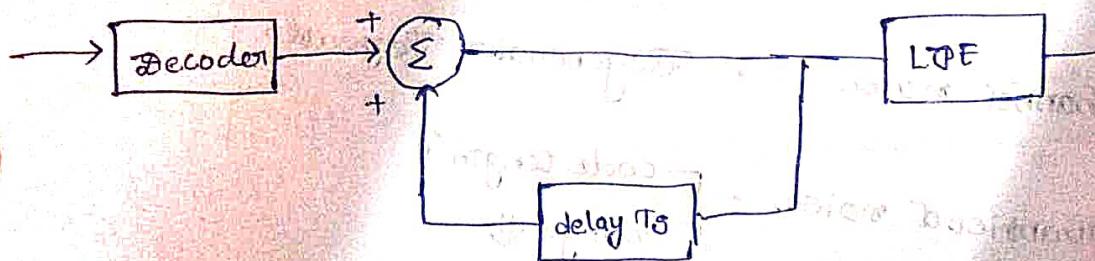
$$x_e(t) = M(t) - M(t-T_s)$$

+ve then 1
-ve then 0

* One bit per sample

* Constant Slope.

Demodulation of DM

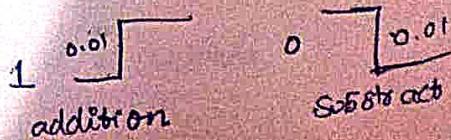


$$\text{Step size} = \Delta$$

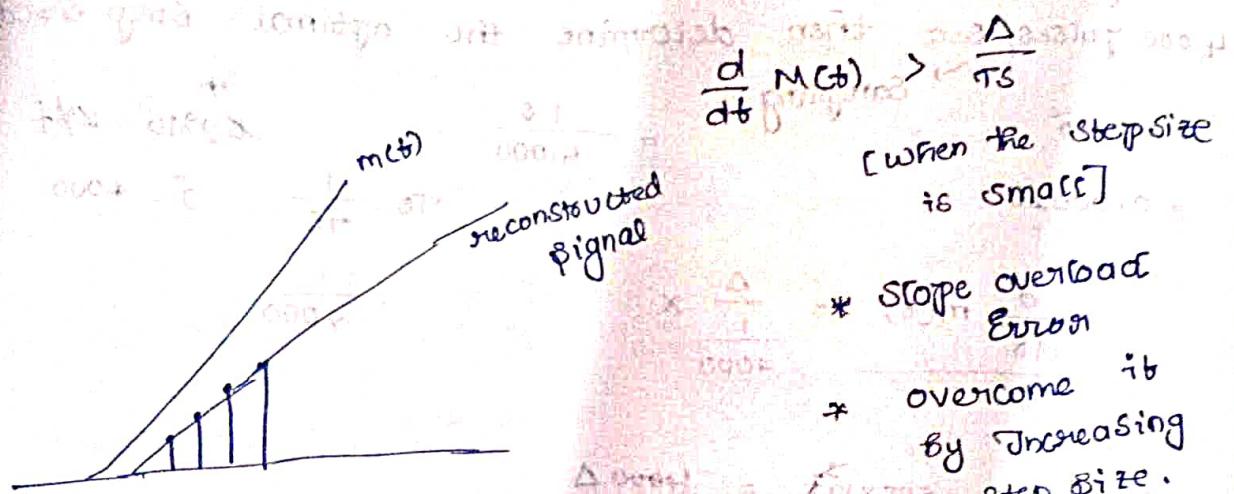
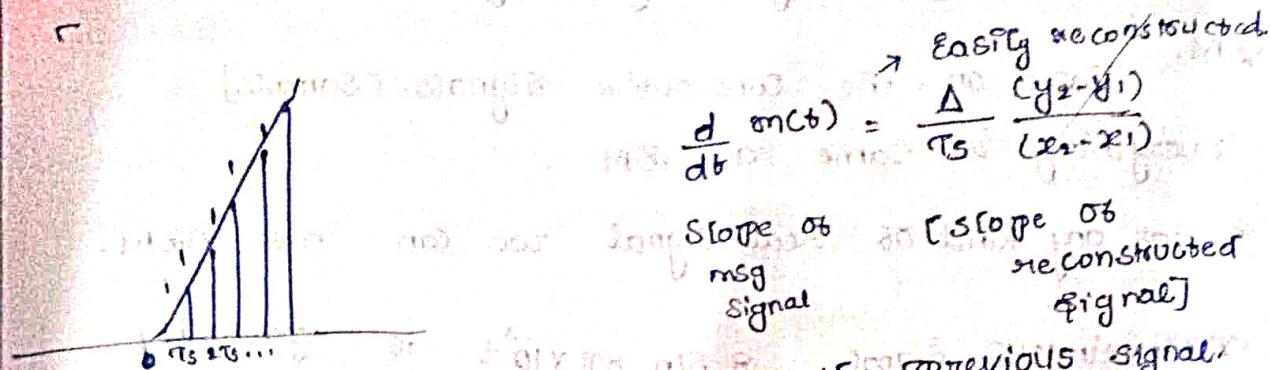
If it receive 1 then Δ is added to previous value.

it receive 0 then Δ is subtracted from previous value.

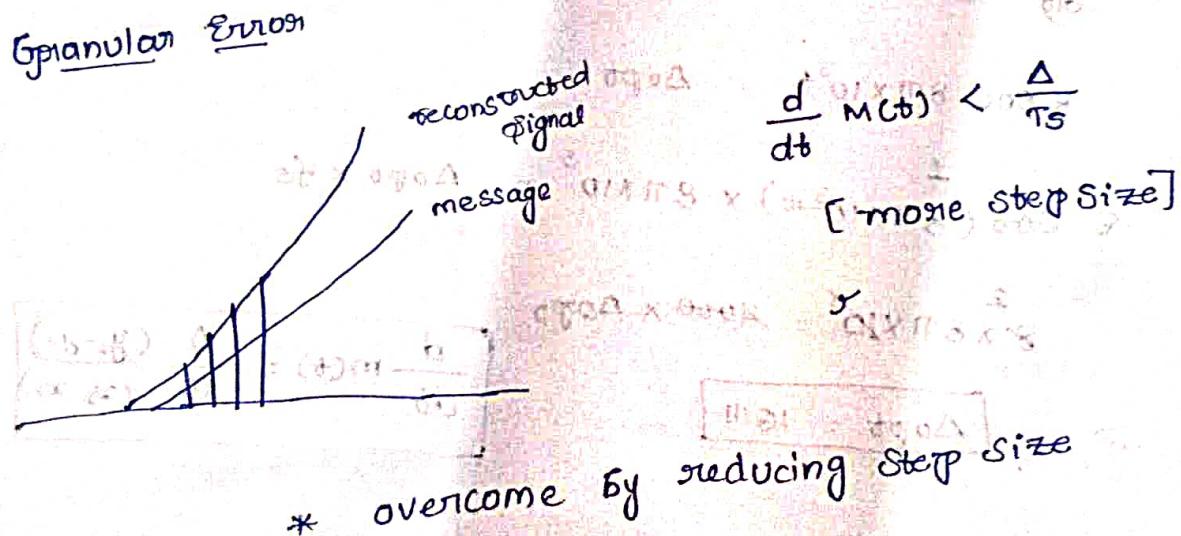
Previous value



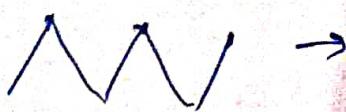
Slope is same then we can get the same signal.



- * Slope overload Error
- * Overcome it by Increasing step size.



can't construct as step size is not constant (or) slope is not constant
[Adaptive Delta Modulation]



constant we can use delta modulation.

AZDM \rightarrow Step size may vary only difference from DDM.

Based on the consecutive signals. [sample]
Everything is same as DDM.

* for any kind of real signal we can use AZDM.

A continuous signal $8 \sin 8\pi \times 10^3 t$ is passed through Delta Modulation whose pulse rate is 4,000 pulses/sec then determine the optimal step size.

$$\begin{aligned} &= 0.25 \times 10^{-3} \\ &= 0.00025 \quad \text{Sampling} = \frac{1}{4,000} \quad 8\pi \times 10^3 = 2\pi f \\ &\quad T_S = \frac{1}{f_S} \quad f = 4000 \end{aligned}$$

$$\frac{d}{dt} m(t) = \frac{\Delta}{4000} \times \frac{1}{4000}$$

$$8 \frac{d}{dt} \sin 8\pi \times 10^3 t = 4000 \Delta$$

$$8 \cos 8\pi \times 10^3 t = \Delta_{opt}$$

$$8 \cos(8\pi \times 10^3 t) \times 8\pi \times 10^3 = \Delta_{opt} \times f_S$$

$$8 \times 8\pi \times 10^3 = 4000 \times \Delta_{opt}$$

$$\boxed{\Delta_{opt} = 16\pi}$$

$$\boxed{\frac{d}{dt} m(t) = \frac{\Delta}{T_S} \frac{(y_2 - y_1)}{(x_2 - x_1)}}$$

$$\frac{d}{dt} m(t) = \frac{\Delta}{T_S} \frac{(y_2 - y_1)}{(x_2 - x_1)} > \frac{\Delta}{T_S} \Rightarrow \text{granular error}$$

$$\frac{d}{dt} m(t) = \frac{\Delta}{T_S} \frac{(y_2 - y_1)}{(x_2 - x_1)} < \frac{\Delta}{T_S} \Rightarrow \text{overflow error.}$$

A Sinusoidal msg signal, of frequency fm and amplitude Am is passed through delta Modulation whose step size is 0.628 volt. Given Sampling rate is 4000 samples/sec. For which of the following delta modulation will stoppe overload Error.

$$\frac{d}{dt} m(t) = 0.628$$

$$2\pi(1 \times 10^3) = \omega$$

$$2\pi f = 2000\pi$$

1. $Am = 3V$ $fm = 1K$ ✓
2. $Am = 2V$ $fm = 1.5K$
3. $Am = 2V$ $fm = 2.5K$
- ④ $Am = 1V$ $fm = 2.5K$

$$\frac{d}{dt} m(t) \geq \frac{\Delta}{T_s} \frac{(y_2 - y_1)}{(x_2 - x_1)} \geq \frac{\Delta}{T_s} \frac{0.628 \times 4000}{2\pi} = \frac{0.628 \times 4000}{2\pi}$$

$$3 \sin 2\pi \times 10^3 t = \frac{\Delta opt \times 4000}{2\pi(1.5)}$$

$$3 \sin 1\pi \times 10^3 t \times 2\pi \times 10^3 t = \Delta opt \times 4000$$

$$3\pi = \Delta opt \times 4$$

$$\Delta opt = 2.356 < 2512$$

$$4.712 > 0.628$$

$$\Rightarrow 2 \sin 3\pi \times 10^3 t = \Delta opt \times 4000$$

$$18.72 > 0.628$$

$$2 \cos 3\pi \times 10^3 t \times \sin 3\pi \times 10^3 t = \Delta opt \times 4000$$

$$31.41 > 0.628$$

$$15.73 >$$

$$fm = 1000$$

$$6\pi \times 10^3 = \Delta opt \times 4000$$

$$Am \sin \omega t$$

$$\Delta opt = 4.712$$

$$\omega = 2\pi f_m \\ = 2\pi (1 \times 10^3)$$

$$\omega = 2\pi \times 10^3$$

$$6\pi > 0.628$$

$$(2) 6\pi > 0.628$$

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

$$3. \frac{d}{dt} 2 \sin 5\pi \times 10^3 t > 0.628 \times 4000$$

$$2 \cos 5\pi \times 10^3 t \times 5\pi \times 10^3 > 0.628 \times 4000$$

$$10\pi \times 10^3 > 0.628 \times 4000$$

$$10\pi > 0.628 \times 4$$

$$31.41 > 2.512 \quad \checkmark$$

$$4. \frac{d}{dt} 1 \cos 5\pi \times 10^3 t > 0.628 \times 4000$$

$$= 5\pi \times 10^3 > 0.628 \times 4000$$

$$5\pi > 2.512$$

$$15.7 > 2.512 \quad \checkmark$$

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

$$\Delta = 0.628 \text{ Nm per degree}$$

$$f_S = 4000$$

$$T_S = \frac{1}{f_S}$$

$$1. A_m = 3V \quad f_m = 1K$$

$$\frac{d}{dt} 3 \sin 2\pi \times 10^3 t > 0.628 \times 4000$$

$$3 \boxed{\cos 2\pi \times 10^3 t} \times 2\pi \times 10^3 > 2.512 \times 1000$$

$$3 \times 2\pi \times 10^3 > 2.512 \times 1000$$

$$6\pi > 2.512$$

$$4. 512 > 2.512 \text{ (Overload error)}$$

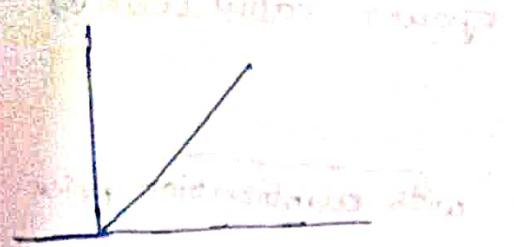
$$r(t) = 125 \{ u(t) - u(t-1) \} + (250 - 125t) \{ u(t-1) - u(t-2) \}$$

From the following msg signals, the optimal step size with Sampling Frequency as 1000 samples/sec.

$$f_s = 1000$$

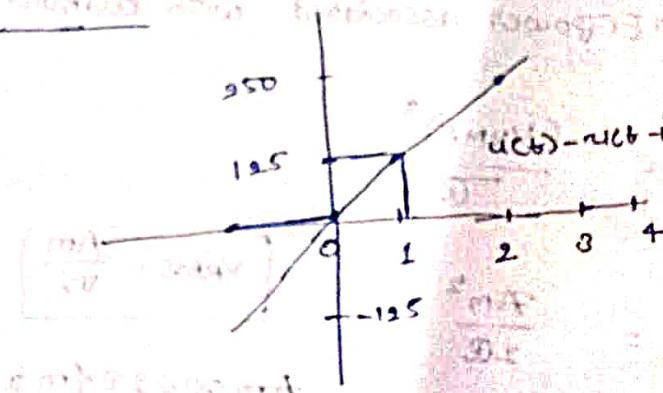
$$t = \frac{1}{1000} \quad t = 10^{-3}$$

$$= (125t) \{ u(t) - u(t-1) \} + (250 - 125t) \{ u(t-1) - u(t-2) \}$$



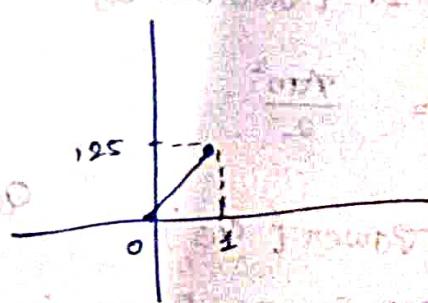
$$r(t) = 125t \quad t > 0$$

otherwise

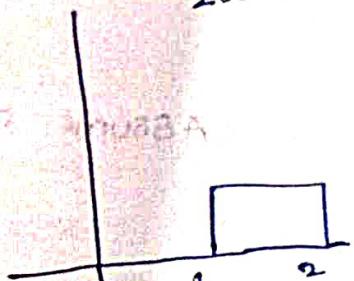


$$125t$$

$$250 - 125t$$

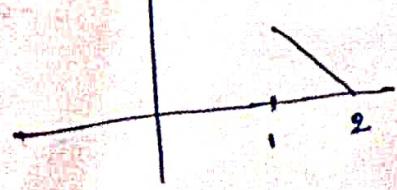
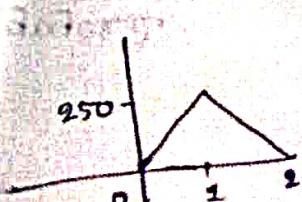
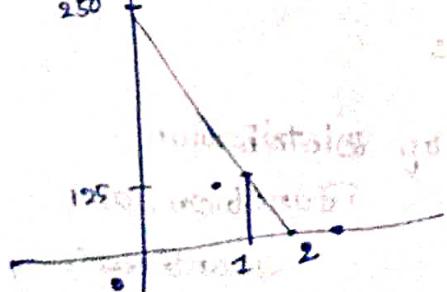


$$[125t] - u(t) - u(t-1)$$



$$\Rightarrow u(t-1) - u(t-2)$$

$$+ 250 - 125(t) \times \\ u(t-1) \times u(t-2)$$



$$\frac{d}{dt} m(t) > \frac{\Delta}{T_s} \times 125$$

$$\frac{d}{dt} r(t) > \frac{\Delta}{T_s} \times 125 \times 1000$$

Slope of msg. > $\frac{\Delta}{T_s}$
Signal

$$125 > \Delta \times 1000$$

$$\boxed{\Delta_{opt} = 0.125}$$

Signal to Quantization noise power ratio [SQNR]

$$SQNR = \frac{\text{Signal power}}{\text{Noise power} [\text{power associated with quantization noise}]}$$

$$\text{Signal power} = \frac{(V_{RMS})^2}{R}$$

$$= \frac{A_m^2}{2R}$$

$$V_{RMS} = \frac{A_m}{\sqrt{2}}$$

$$A_m \cos 2\pi f_m t$$

Assume $R = 1$. [Resistance]

$$= \frac{A_m^2}{2}$$

Noise Power: $\{\text{Power } \{Q_e\}\}$ $Q_e \rightarrow \text{Quantization}$
Error = random number.

Mean Square Error to get Power.

$$= \text{MSE } \{Q_e\}$$

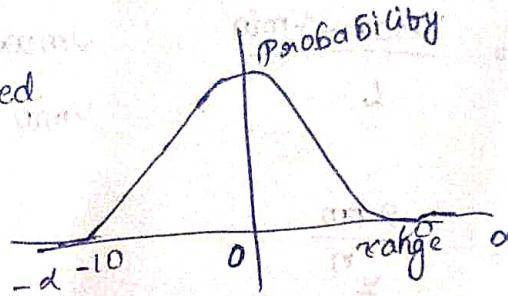
$$= \int_{-\alpha}^{\alpha} Q_e^2 F(Q_e) dQ_e$$

Probability Distribution
Function of
Quantization.

Probability

Random number is generated near to zero itself.

$$\int_{-\infty}^{\infty} f(x) dx = 1$$



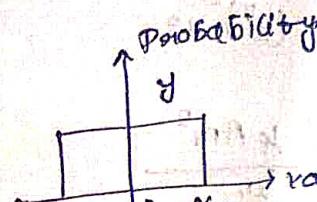
Gaussian PDF

[Bell shape PDF]

In that range we will get

equal probability.

[uniform distribution]



which range
the random number
distributed.

Range of Quantization error is $\pm \frac{\Delta}{2}$ to $\frac{\Delta}{2}$

Magnitude is $1/\Delta$ which gives us area = 1

$$= \int_{-\frac{\Delta}{2}}^{\frac{\Delta}{2}} \frac{1}{\Delta} Qe^2 dQe \quad F(Qe) = \frac{1}{\Delta}$$

$$= \frac{1}{\Delta} \left[\frac{Qe^3}{3} \right]_{-\frac{\Delta}{2}}^{\frac{\Delta}{2}}$$

$$= \frac{1}{\Delta} \times \frac{1}{3} \left[\left(\frac{\Delta}{2} \right)^3 - \left(-\frac{\Delta}{2} \right)^3 \right]$$

$$= \frac{1}{\Delta} \times \frac{1}{3} \times \frac{\Delta^3}{2} + \frac{\Delta^3}{2} \Rightarrow \frac{1}{\Delta} \times \frac{\Delta^3}{24} + \frac{\Delta^3}{24}$$

$$= \frac{2}{34} \times \frac{\Delta^3}{8+2} = \frac{2}{34} \times \frac{\Delta^3}{10}$$

$$= \frac{\Delta^2}{6}$$

$$= \frac{1}{\Delta} \times \frac{2\Delta^3}{24+12} = \frac{\Delta^2}{12}$$

$$\Delta = \frac{V_{max} - V_{min}}{L} \quad V_{max} = A_m \\ V_{min} = -A_m$$

$$\Delta = \frac{2A_m}{2^n}$$

$$= \frac{(2A_m)^2}{(2^n)^2 / 12}$$

$$= \frac{4A_m^2}{2^{2n} / 12}$$

$$= \frac{A_m^2}{3 \cdot 2^{2n}} \quad [\text{Noise power}]$$

$$SQNR = \frac{\frac{A_m^2}{8}}{\frac{A_m^2}{3 \cdot 2^{2n}}} = \frac{3 \cdot 2^{2n}}{8} = \frac{3}{2} \cdot 2^{2n}$$

Signal Power
Noise Power

$$SQNR = (1.5) 2^{2n}$$

n	
1	6
2	24
3	96
4	384
5	1536

$$\frac{18 \times \frac{3}{2}}{89 \times \frac{3}{2}} = \frac{1}{4}$$

$$(SQNR)_2 = 4 (SQNR)_1 2^2$$

$$(SQNR)_3 = 4 (SQNR)_2 2^2$$

$$(SQNR)_4 = 4 (SQNR)_3 2^2$$

$$(SQNR)_5 = 4 (SQNR)_4 2^2$$

Thus $n \rightarrow n+k$, The SQNR increases

$$1+4=5$$

By $\frac{2^{2k}}{2}$

SQNR in terms of dB.

$$= 10 \log_{10}(\text{SQNR})$$

$$= 10 \log_{10} \left[\frac{3}{2} \cdot 2^{2n} \right]$$

$$= 2n \times 10 \log_{10} [3]$$

$$= 1.76 + 6.02 n$$

n	$1.76 + 6.02 n$	$\frac{12.06}{1.76}$	13.82
1	7.78		
2	13.82		
3	19.82		
4	25.84		
5	31.86		
6	37.88		

Has $n \rightarrow n+k$ The SQNR increase in decibel then

20dB then k Bits.

A msg signal of $8 \sin 8\pi \times 10^8 t$ is to be transmitted through PCM where sampling rate is 50% higher than nyquist rate. The minimum SQNR required for transmission is 22 dB, determine transmission bandwidth required. SQNR in dB.

$$n=4 \quad 2^4 = 16$$

$$f_S = \frac{1}{T_S}$$

$$f_S > 2f_m$$

$$[8 \times 10^8]$$

$$8 \times 10^8 = 2f_m$$

$$T_S = \frac{1}{f_S}$$

50% Higher than 8000

12000

$$8000 \times \frac{50}{100} = 4000$$

$$\frac{1}{12000} = \frac{1}{4}$$

$$B.D. = \frac{1}{48000}$$

$8 \Rightarrow 16000$

Bit rate = 48000

A msg signal of peak to peak voltage 1.536 Volts is passed through PCM system having 12.8 quantisation levels. Find Quantisation noise power

$$= \frac{A_m^2}{3 \cdot 2^{2n}}$$

$n =$

$$2A_m = 1.536$$

$$A_m = \frac{1.536}{2}$$

$$= \frac{(1.536)^2}{3 \cdot 2^{14}}$$

$$= \frac{1}{3} \times \frac{(1.536)^2}{2^{14}}$$

$$= \frac{1}{3} \times \frac{2.359}{16384}$$

$$= \frac{2.359}{49152}$$

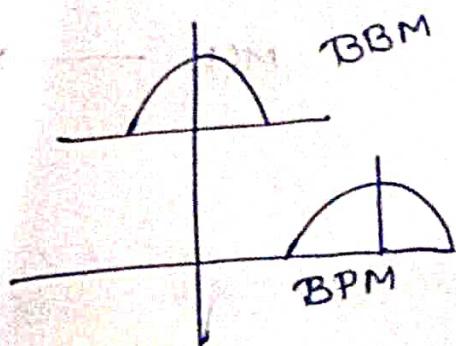
$$= 1.2 \times 10^{-5}$$

Band pass Modulation

After modulation \rightarrow BPM

Before modulation (or) msg signal is called **Base Band Modulation.**

* Band pass for longer distance
(or) wireless signal transmission.



Binary Signaling
(Single - Bit transmission)

M-array signaling.

[symbols - group of bits]

2-Bits, 4-Bits, 8-Bits
(or) 16-Bits

ASK PSK FSK
Amplitude shift keying Phase shift keying Frequency shift keying

ASK - [on/off keying]

$1 \rightarrow \text{ACOS}2\pi fct$

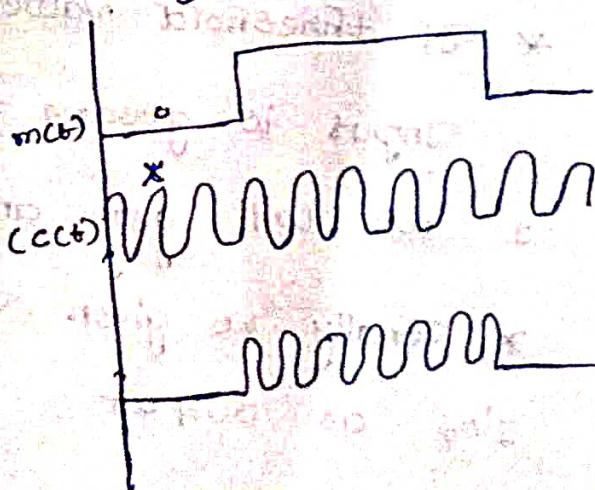
$0 \rightarrow \text{No Signal}$ [zero transmission (or) no signal]

From zero transmission we
don't transmit any kind of

signal

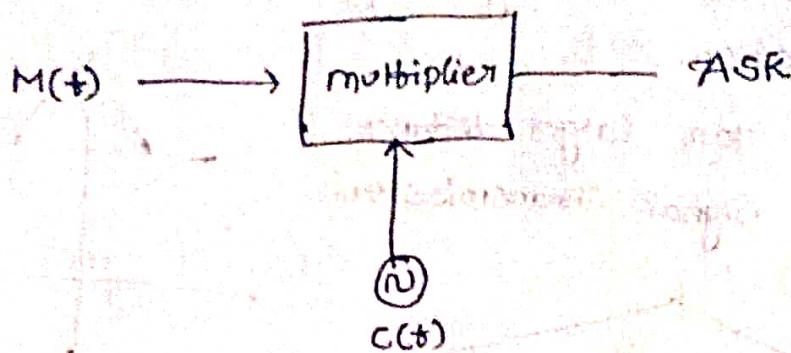
$$0 \times \text{anything} = 0$$

$$1 \times \text{anything} = \text{anything}$$

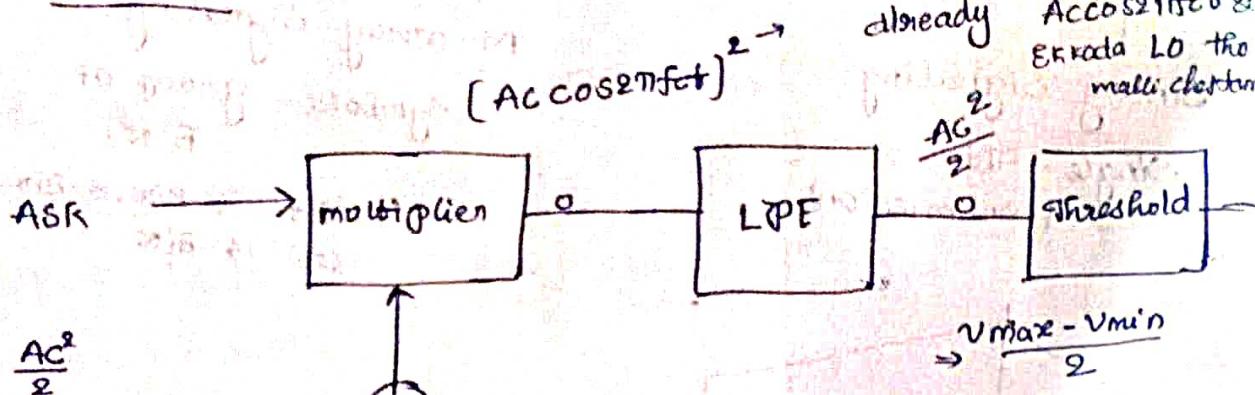


Phase Shift Keying

Modulation



De-Modulation



$$AC^2 \cos^2 2\pi f_{\text{ct}} t$$

$$AC^2 \left(1 + \frac{\cos 2\pi f_{\text{ct}} t}{2} \right)$$

$$\frac{AC^2}{2} + \frac{AC^2 \cos 2\pi f_{\text{ct}} t}{2}$$

$$\text{LPF} \rightarrow \frac{AC^2}{2} = V_{\text{max}}$$

* If threshold value is greater than ASK, it is 0

Input is greater than threshold value then

1 we will get 1 and vice-versa is 0.

* Finally we given Input as 1 & 0's, we get 1's & 0's as output

$$(\text{Acc} \cos 2\pi f_{\text{ct}} t). A_c \cos(2\pi f_{\text{ct}} t + \phi) t$$

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

$$= \frac{A_c^2 \cos^2 \phi}{2}$$

$$\cos A \cdot \cos B$$

$$\cos(A+B) + \cos(A-B)$$

\Rightarrow It is quadrature null effect.

Energy for Bit

1 → Carrier Signal. (Energy required for transmitting 1]

$$0 \rightarrow \int_0^{T_B} (0) dt = 0$$

$$1 \rightarrow \int_0^{T_B} (\text{Acc} \cos 2\pi f_{\text{ct}} t)^2 dt$$

$$= \int_0^{T_B} \frac{A_c^2}{2} + \frac{A_c^2 \cos 2\pi f_{\text{ct}} t}{2} dt$$

$$\int \cos t = -\sin t$$

$$t = \frac{t^2}{2}$$

$$= \frac{A_c^2}{2} \int_0^{T_B} \frac{A_c^2 \cos 2\pi f_{\text{ct}} t}{2} dt$$

$$\text{Acc} \\ \cos^2 \theta = 1 + \frac{\cos 2\theta}{2}$$

$$= \frac{A_c^2}{2} \int_0^{T_B} [1 + \cos 2\pi f_{\text{ct}} t] dt$$

$$= \frac{A_c^2}{2} \left[[t + \frac{\sin 2\pi f_{\text{ct}} t}{2}] \right]_0^{T_B}$$

Integrate a sin signal over one period is zero.

$$= \frac{A_c^2}{2} T_B + \cancel{\frac{A_c^2}{2} \sin 2\pi f_{\text{ct}} t B}_0$$

$$E_B = \boxed{\frac{A_c^2}{2} T_B}$$

$$1 \rightarrow \frac{A_c^2}{2} T_B \quad \left. \begin{array}{l} \\ \end{array} \right\} \text{Energy required}$$

$$I \rightarrow A_c \cos 2\pi f_c t$$

$$E = \frac{A_c^2 T_B}{2}$$

$$I \rightarrow \sqrt{\frac{2EB}{TB}} \cos 2\pi f_c t$$

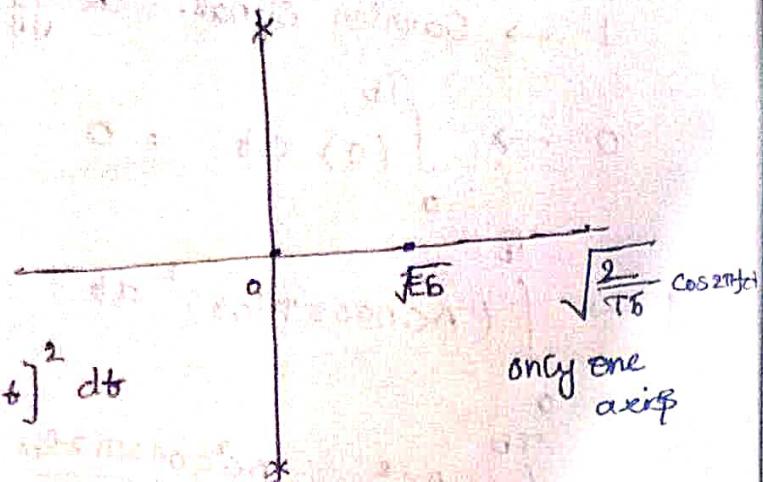
$$A_c^2 = \frac{2EB}{TB}$$

$$A_c = \sqrt{\frac{2EB}{TB}}$$

Constellation Diagram

$$I \rightarrow \sqrt{\frac{2EB}{TB}} \cos 2\pi f_c t$$

$$0 \rightarrow 0$$



$$EB = \int_0^{T_B} \left(\sqrt{\frac{2EB}{TB}} \cos 2\pi f_c t \right)^2 dt$$

$E = 1 \Rightarrow$ Normalised Functions

$$EB = EB \int_0^{T_B} \left(\sqrt{\frac{2}{TB}} \cos 2\pi f_c t \right)^2 dt$$

we will get and called as Normalised Function.
Binary ASK, where one axis and two points 0 & 1

Energy from Constellation diagram

Energy = (Distance from original) \times initial.

$$E_0 = 0$$

$$E_1 = (\sqrt{EB})^2 = EB$$

Probability of Error = distance b/w two symbols

giving 1 getting 0 = $\sqrt{2B} - 0$

is error.

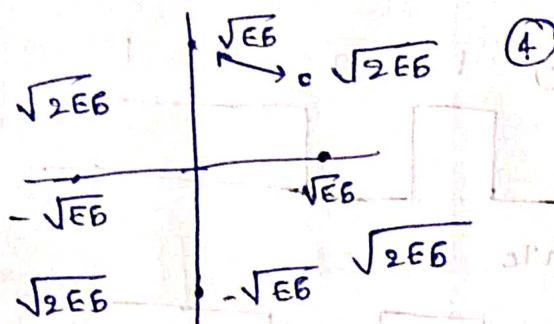
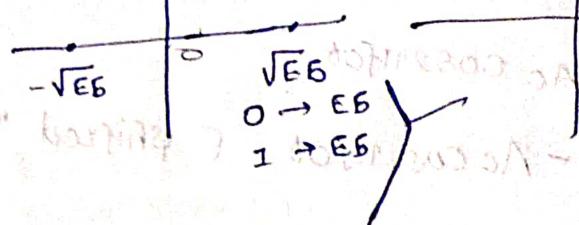
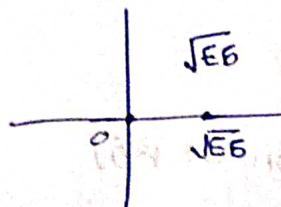
$$= \sqrt{2B}$$

Probability of Error giving 1 getting 0 is error.

$$= \sqrt{E_B} - 0$$

$$\sqrt{E_B}$$

$$\sqrt{E_B} \sqrt{2E_B}$$



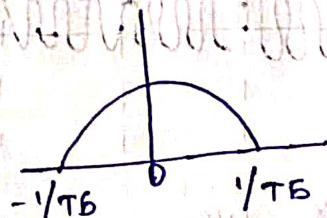
more distance is, 2 then T_b has least probability of error.

④ has the least at that case [less distance].

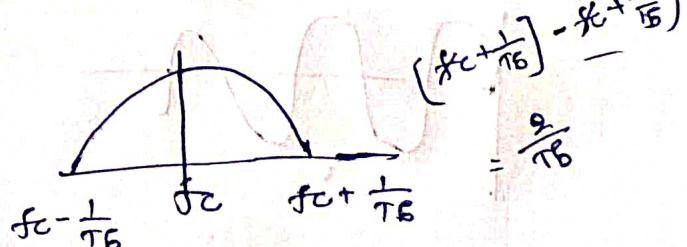
and it has more probability of error,

Energy then $[\sqrt{E_B}]^2 \rightarrow (\text{distance from origin})^2$

Bandwidth of ASK



ASK



$$B.W = 2f_C \times \frac{1}{T_B}$$

$$B.W = \frac{2 \times 1}{T_B} = 2 T_B \quad [\text{twice of bit rate}]$$

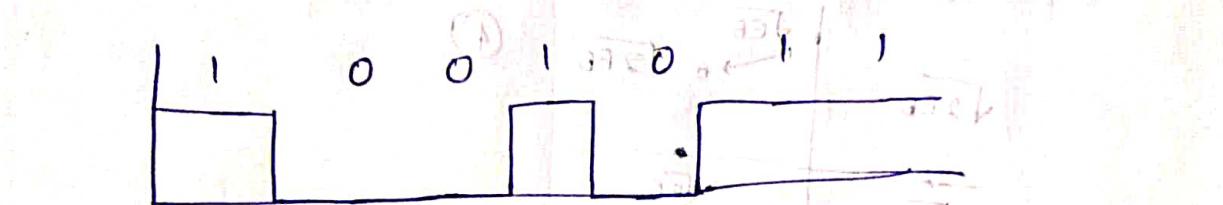
DPSK: Phase Shift Keying

Binary DPSK 360° But 2 then

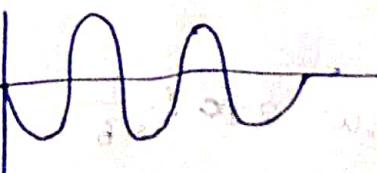
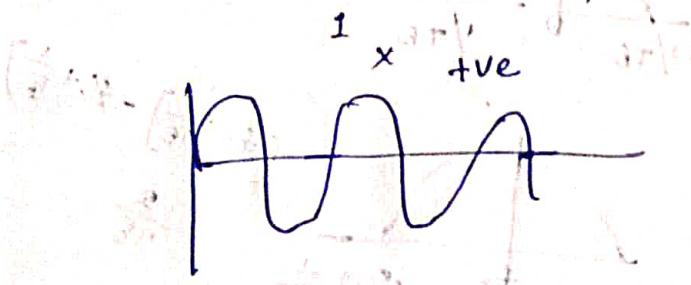
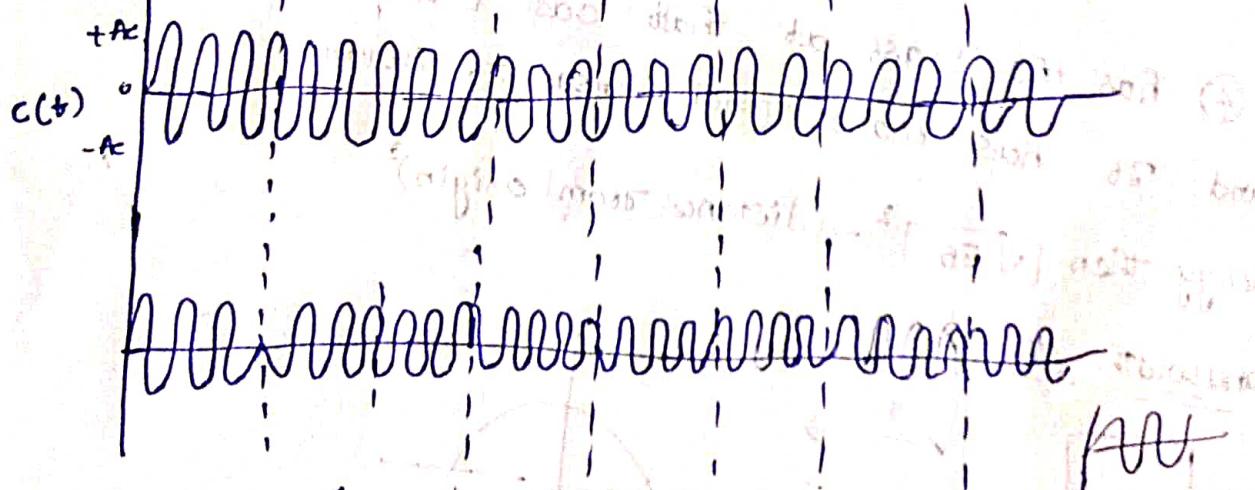
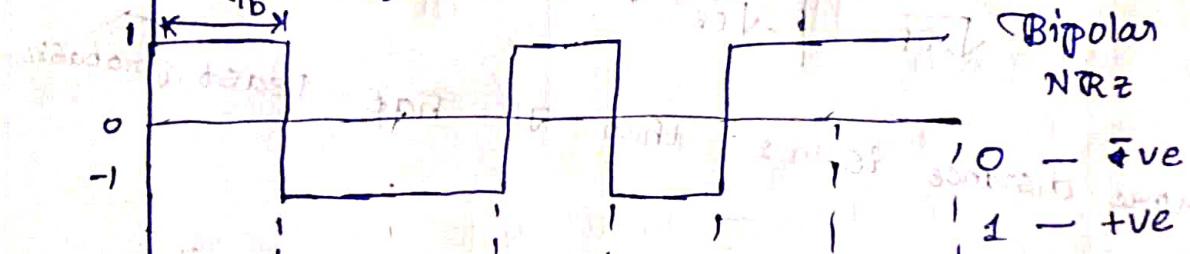
S-PSK: $\frac{360^\circ}{2} = 180^\circ$

$1 \rightarrow Ac \cos 2\pi fct$

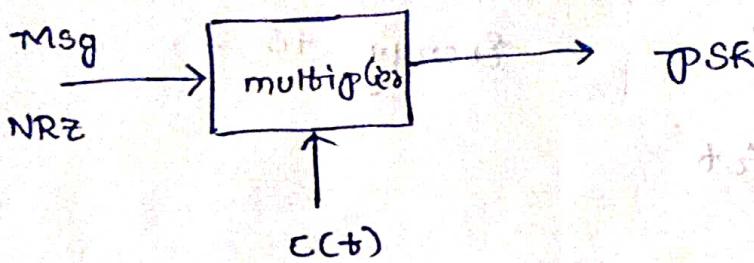
$0 \rightarrow -Ac \cos 2\pi fct$ [shifted version of 180°]



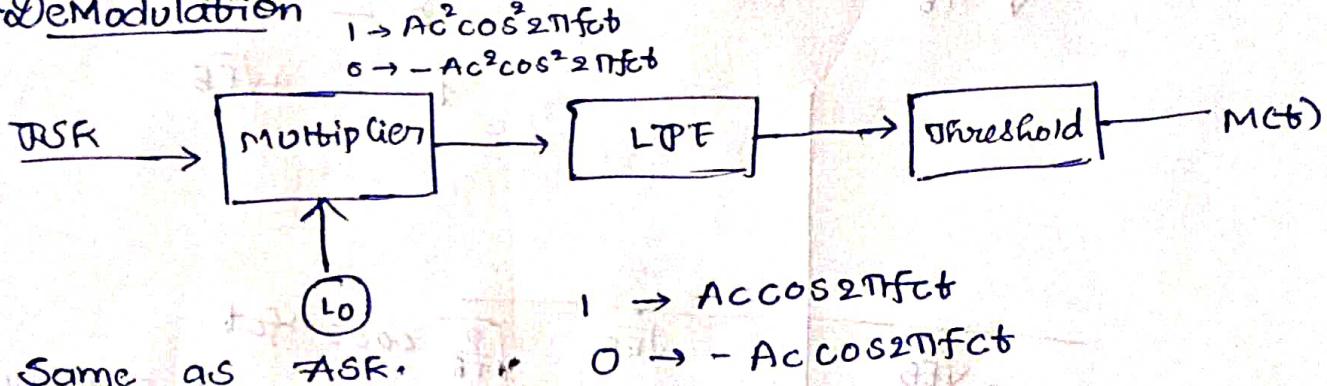
$$T_B = n T_C$$



Modulation of PSK



DeModulation



Multiplier

$$1 \rightarrow \frac{AC^2}{2} + \frac{AC^2}{2} \cos(2\pi f_c t)$$

$$0 \rightarrow -\frac{AC^2}{2} + \frac{AC^2}{2} \cos(-2\pi f_c t)$$

LTF

$$1 \rightarrow \frac{AC^2}{2}$$

$$0 \rightarrow -\frac{AC^2}{2}$$

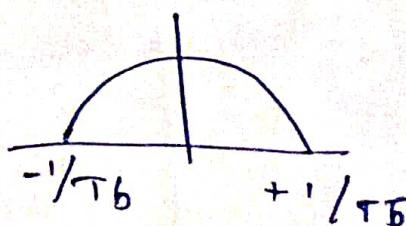
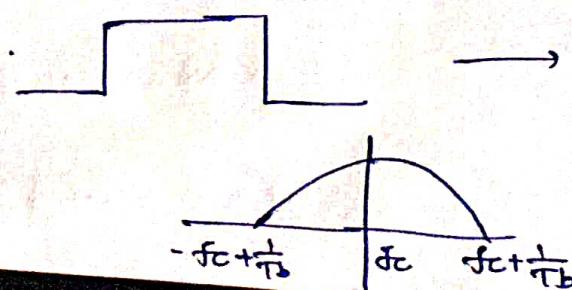
Threshold

$$\frac{v_{max} + v_{min}}{2} = 0 \Rightarrow \frac{\frac{AC^2}{2} - \frac{AC^2}{2}}{2} = 0$$

greater than 0 J_b is 1.

greater less than 0 J_b is 0.

Bandwidth



$$B.W = 2 \times \frac{1}{T_B} = 2f_{RB}$$

Constellation diagram

N.F \Rightarrow any function whose

$$1 \rightarrow A_c \cos 2\pi f_c t$$

Energy is 1.

$$= \sqrt{\frac{2E_B}{T_B}} \cos 2\pi f_c t$$

$$0 \rightarrow -\sqrt{\frac{2E_B}{T_B}} \cos 2\pi f_c t$$

(+)

$$0 \rightarrow -\sqrt{E_B}$$

$$1 \rightarrow +\sqrt{E_B}$$

(+3dB)

black noise

white noise

red noise

black noise

white noise

red noise

$$-\sqrt{E_B}$$

$$\sqrt{E_B}$$

$$\sqrt{\frac{2}{\pi B}} \cos 2\pi f_c t$$

Normalised Function

we will plot y-axis

But won't consider it