

② By using instantaneous sampling, more no. of sigs can be multiplexed since  $T \rightarrow 0$ ,  $N = \frac{T_s}{T}$

### ② Flat-Top Sampling :-

The flat-topped samples constitute the samples in the form of pulses whose top portion is flat & within pulse interval at one

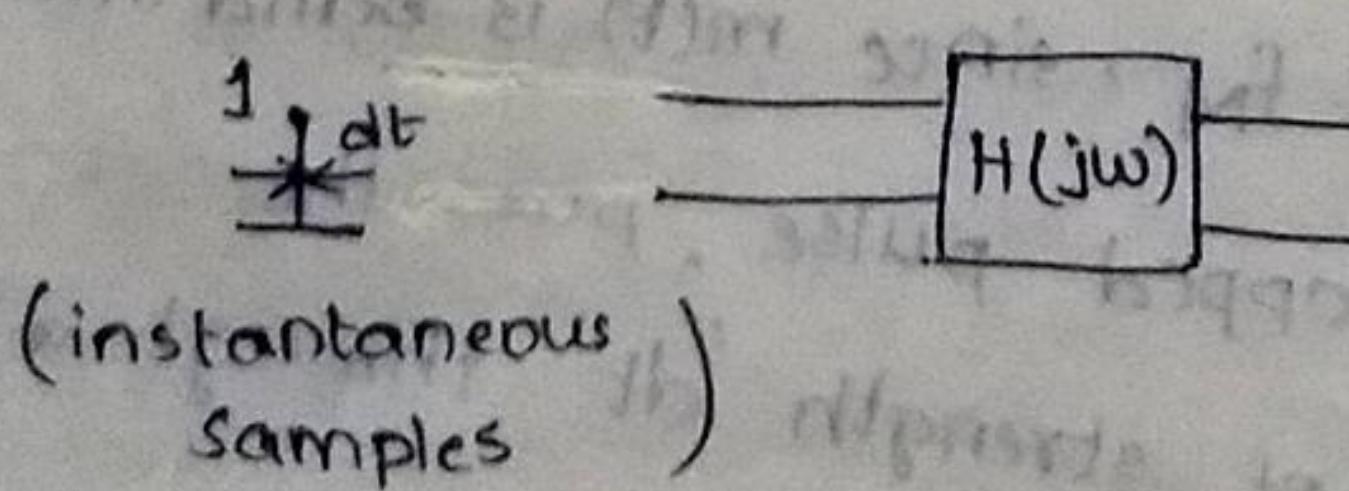
instant of time only the pulse amplitude is equal to the base band sig. amplitude (i.e., it consists of pulses having constant amplitude value established by the sample value of the base band sig. at some point within the pulse interval).

Disadvantage :- In this type of sampling, the base band sig.  $m(t)$  can't be recovered exactly by simply passing the samples through an ideal LPF & at the Rxing end, when these samples are used to reconstruct the sig, more distortion occurs.

### Advantages :-

- 1) Simplifies the design of electronic cktry used to perform the sampling operation.
- 2) Design of flat topped samples is relatively easier & cost of implementation is less.

### Spectrum for sampling methods :-



(instantaneous samples)

↓  
impulse of strength  $dt$

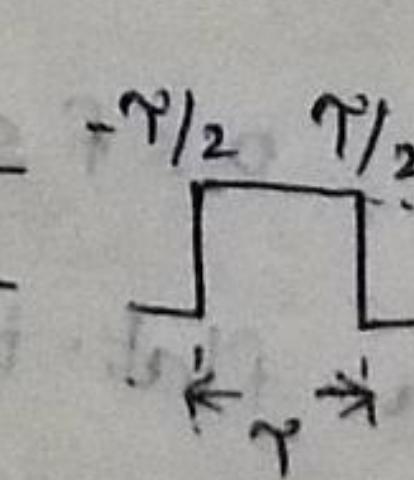


fig :- A n/w with transform  $H(jw)$  which converts a pulse of width  $dt$  into a rectangular pulse of same amplitude, but of duration  $\gamma$ .

$$H(j\omega) \text{ calculation} \quad \left\{ V(f) = F[V(t)] = \int_{-\infty}^{\infty} e^{-j2\pi ft} dt \right\}$$

$$F[\text{impulse of strength } dt \text{ at } t=0] = 1 \times dt$$

$$= dt \downarrow$$

$$F[\text{unit pulse of width } \tau \text{ extending from } -T_1/2 \text{ to } T_1/2] = \frac{\tau}{2} e^{j2\pi f t}$$

$$\text{strength} \quad \left\{ \begin{array}{l} F[\text{impulse}] \\ = \int_{-\infty}^{\infty} e^{-j2\pi ft} dt \Big|_{t=0} \\ = \int_{-\infty}^{\infty} e^0 dt = 1 \end{array} \right.$$

$$= \left[ \frac{e^{-j2\pi f T_1/2}}{-j2\pi f} \right]_{-T_1/2}^{T_1/2}$$

$$= \frac{2j \sin 2\pi f T_1/2}{-j2\pi f}$$

$$\therefore H(j\omega) = \frac{\sin \omega T_1/2 \times \omega T_1/2}{\pi f \times \omega T_1/2}$$

$$= \frac{\sin \omega T_1/2}{\omega T_1/2} \tau$$

$$\therefore H(j\omega) = \frac{F[\text{impulse of strength } dt \text{ at } t=0]}{F[\text{unit pulse of width } \tau \text{ extending from } -T_1/2 \text{ to } T_1/2]}$$

$$H(j\omega) = \frac{F[\text{unit pulse of width } \tau \text{ extending from } -T_1/2 \text{ to } T_1/2]}{F[\text{impulse of strength } dt \text{ at } t=0]}$$

$$\Rightarrow H(j\omega) = \frac{\tau}{dt} \frac{\sin \omega T_1/2}{\omega T_1/2}$$

$$\left\{ H(j\omega) = \frac{F[\text{IP}]}{F[\text{IP}]} \right\}$$

Consider a base band sig whose Fourier transform is  $F[m(t)] = M(j\omega)$ ,  $0 \leq f \leq f_M$ , since  $m(t)$  is band limited to  $f_M$ . To generate a flat-topped pulse, pass the instantaneous sampled sig of strength 'dt' through a filter having transfer fn  $H(j\omega) = \frac{\tau}{dt} \frac{\sin \omega T_1/2}{\omega T_1/2}$ . This  $\eta/\omega$  stretches

impulse of strength  $dt$  to form a unit pulse of width  $\gamma$  sec extending from  $-\frac{\gamma}{2}$  to  $\frac{\gamma}{2}$

$$\therefore F[\text{instantaneous samples}] = F\left[\frac{dt}{T_s}\right] F[m(t)] \\ = \frac{dt}{T_s} M(j\omega) \\ \therefore F[\text{flat topped sampled mlt}] = H(j\omega) F[\text{instantaneous samples}] \\ = \frac{\gamma}{T_s} M(j\omega) \frac{\sin \omega \gamma/2}{\omega \gamma/2}$$

The term  $\frac{\sin \omega \gamma/2}{\omega \gamma/2}$  is termed as aperture factor which is responsible for more distortion in flat topped samples.

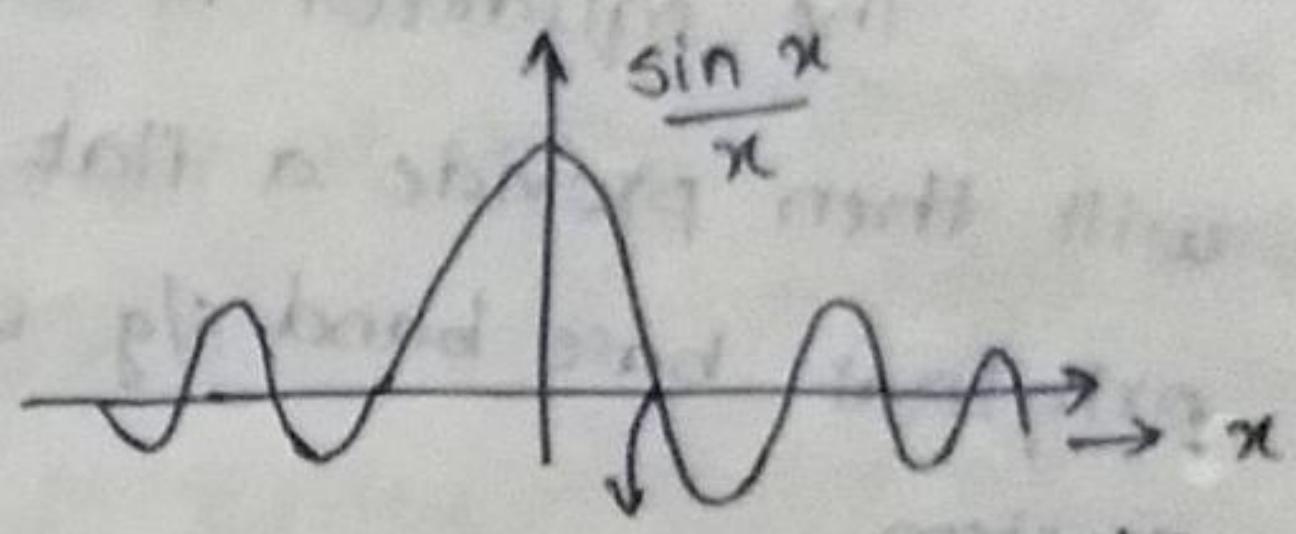
Null freq : The freq at which spectral amplitude becomes zero is called null freq.

Determination of null freq :: ( $f_0$ )

$$\text{At } x = \omega T_s/2 = \pi$$

$$\Rightarrow 2\pi f_0 \frac{\gamma}{2} = \pi$$

$$\Rightarrow f_0 = \frac{1}{\gamma}$$



null points exists at multiples of  $\pi$  i.e.,  $\pi, 2\pi, 3\pi, \dots$

fig a) An idealized spectrum of magnitude, instantaneous base band s/g sampled  $\frac{dt}{T_s} M_0$  mlt)

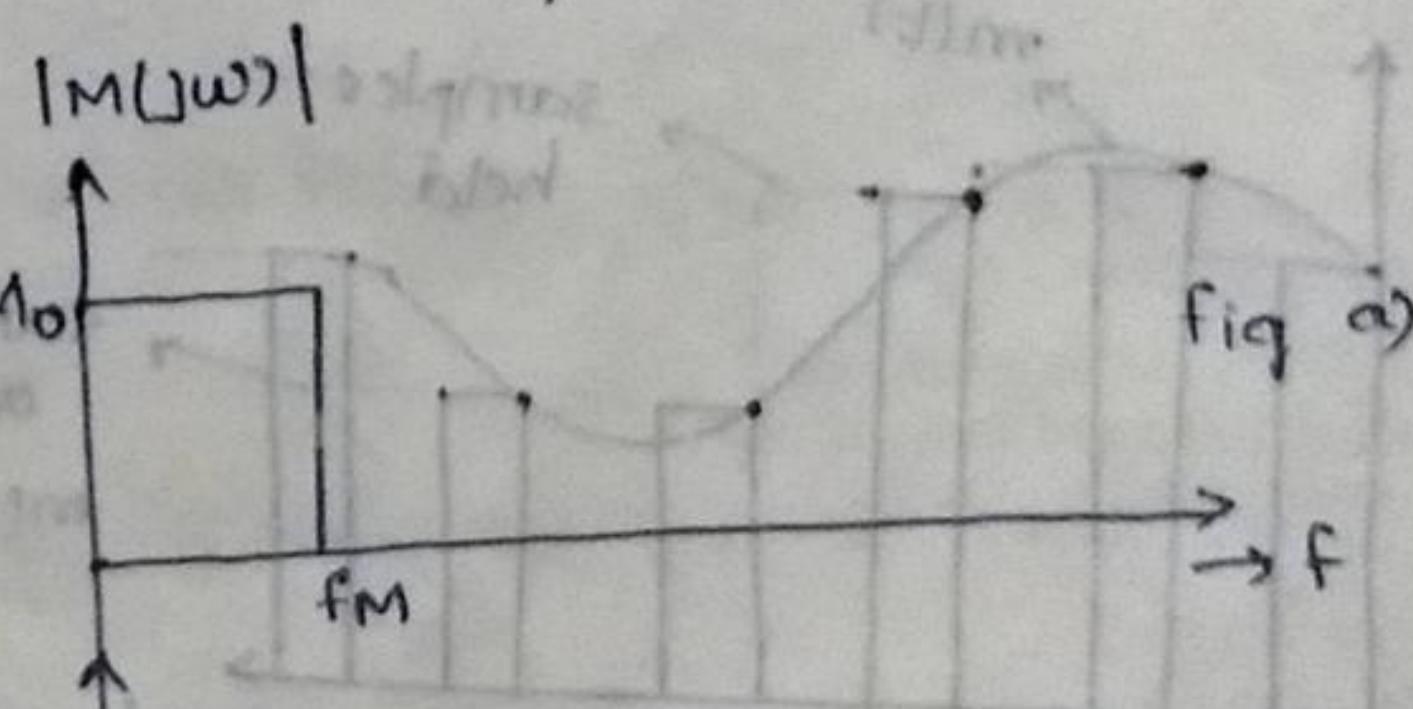


fig b) Spectrum of s/g with instantaneous sampling

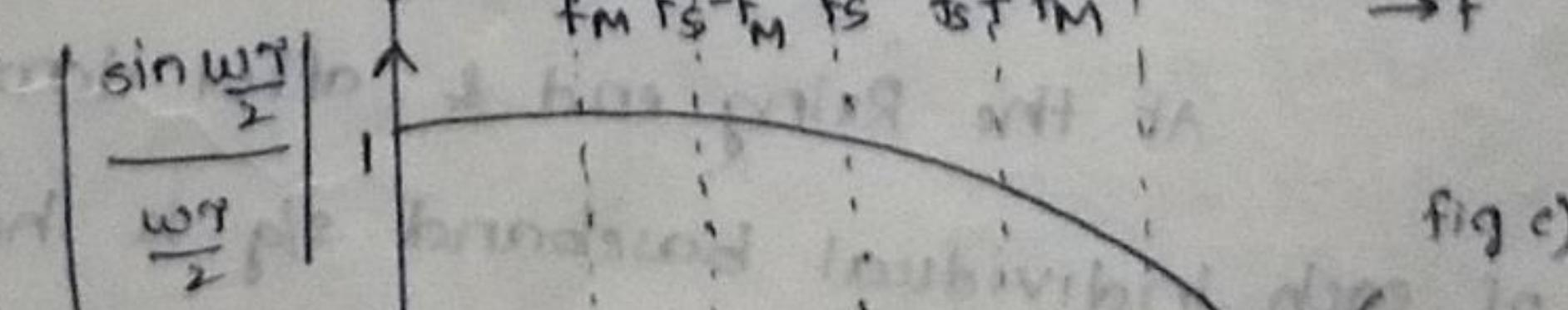


fig c) The form of distortion factor (aperture effect) introduced by flat-topped samples

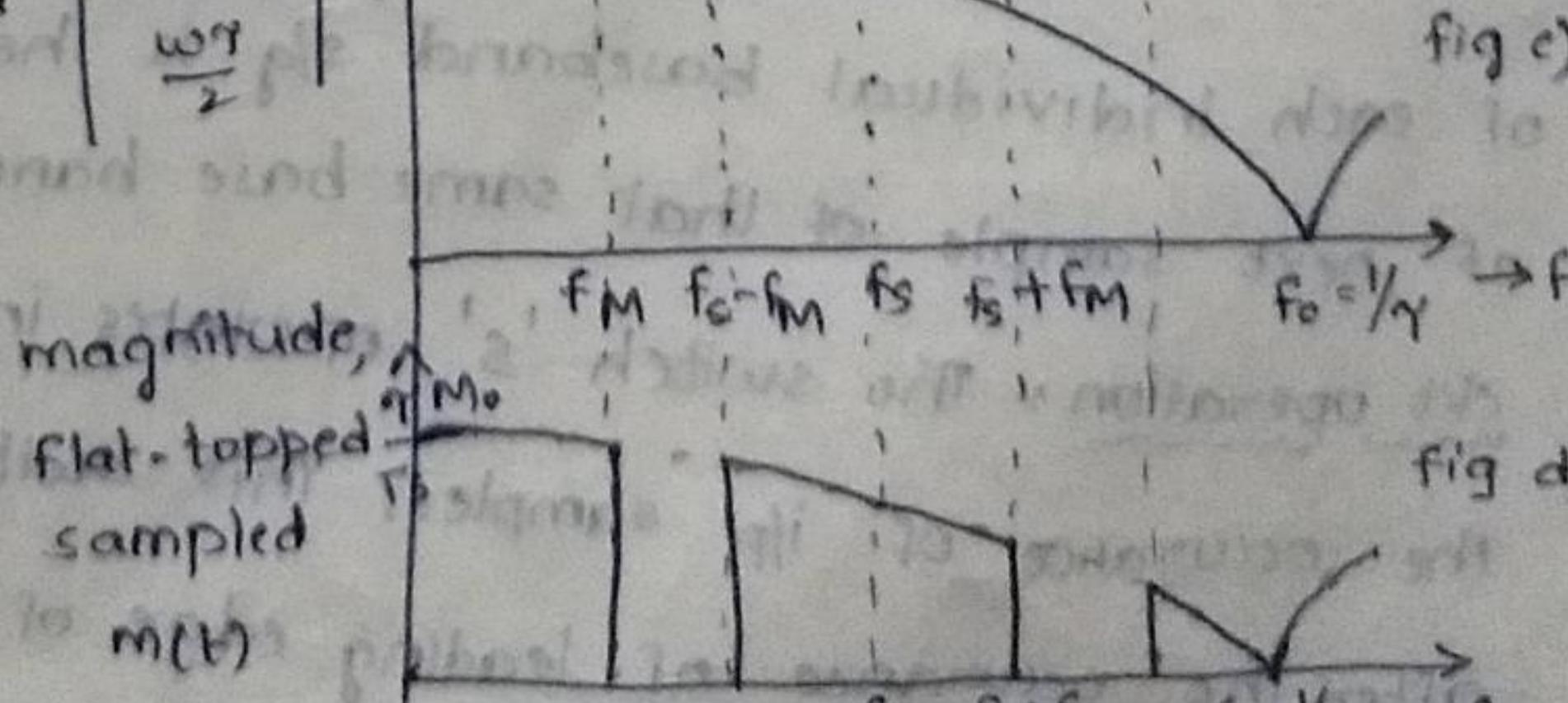
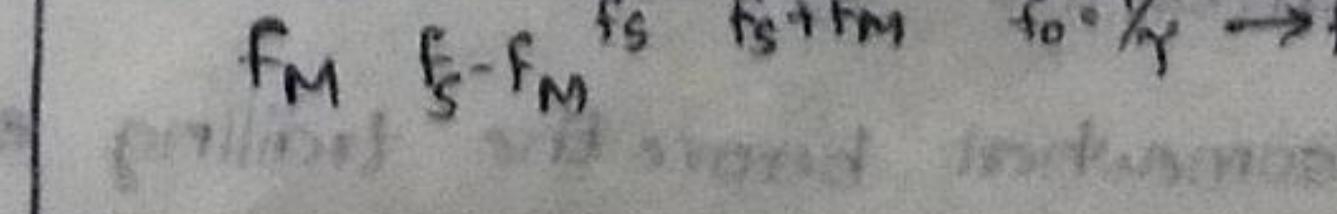


fig d) Spectrum of s/g with flat topped sampling.



The presence of aperture factor is the reason for getting distortion. By choosing null freq  $f_0$  as  $f_0 \gg f_M$ , the distortion can be reduced. But since  $f_0 = \frac{1}{T_s}$ , if  $f_0$  is large,  $T_s$  reduces which reduces the strength of Txed slg.  $\left\{ S_0(t) = \frac{1}{T_s} m(t) \right\}$

### Equalizer :

Alternately pass the slg through an equalizer whose transfer fn. is freq dependent in the form of  $\frac{x}{\sin x}$  and is placed in cascade with o/p LPF.

The equalizer in combination with the aperture effect will then provide a flat overall transfer characteristic b/w original base band slg & the o/p at the Rxing end of the system.

→ Problems (2) [2, 3]

→ signal Recovery through holding :

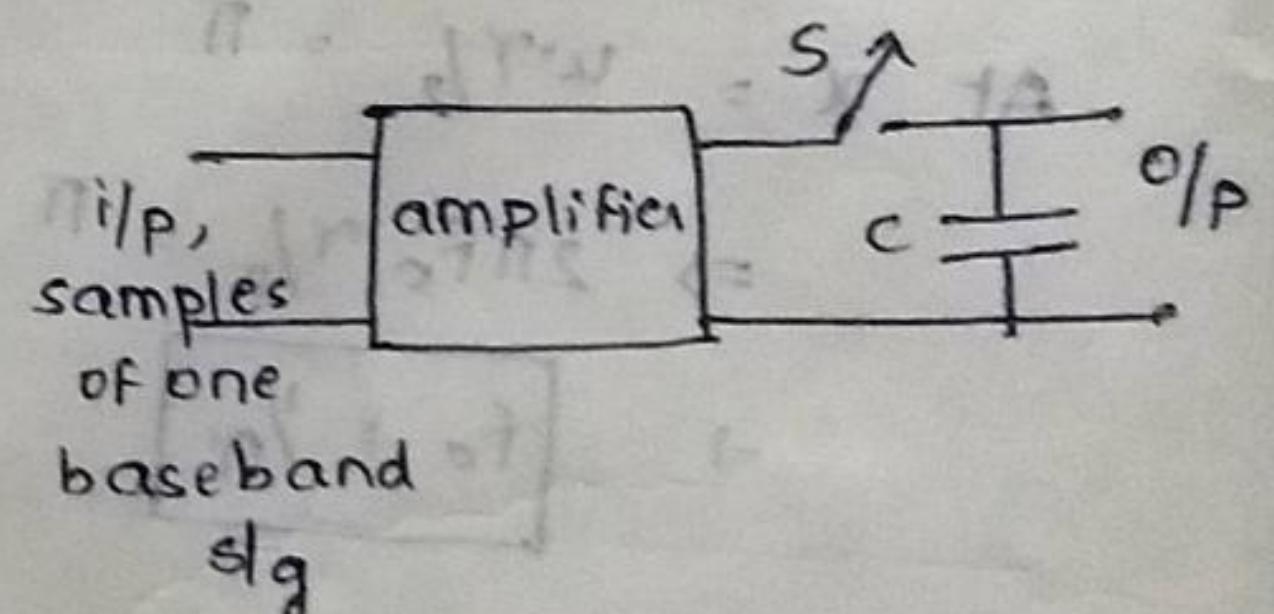
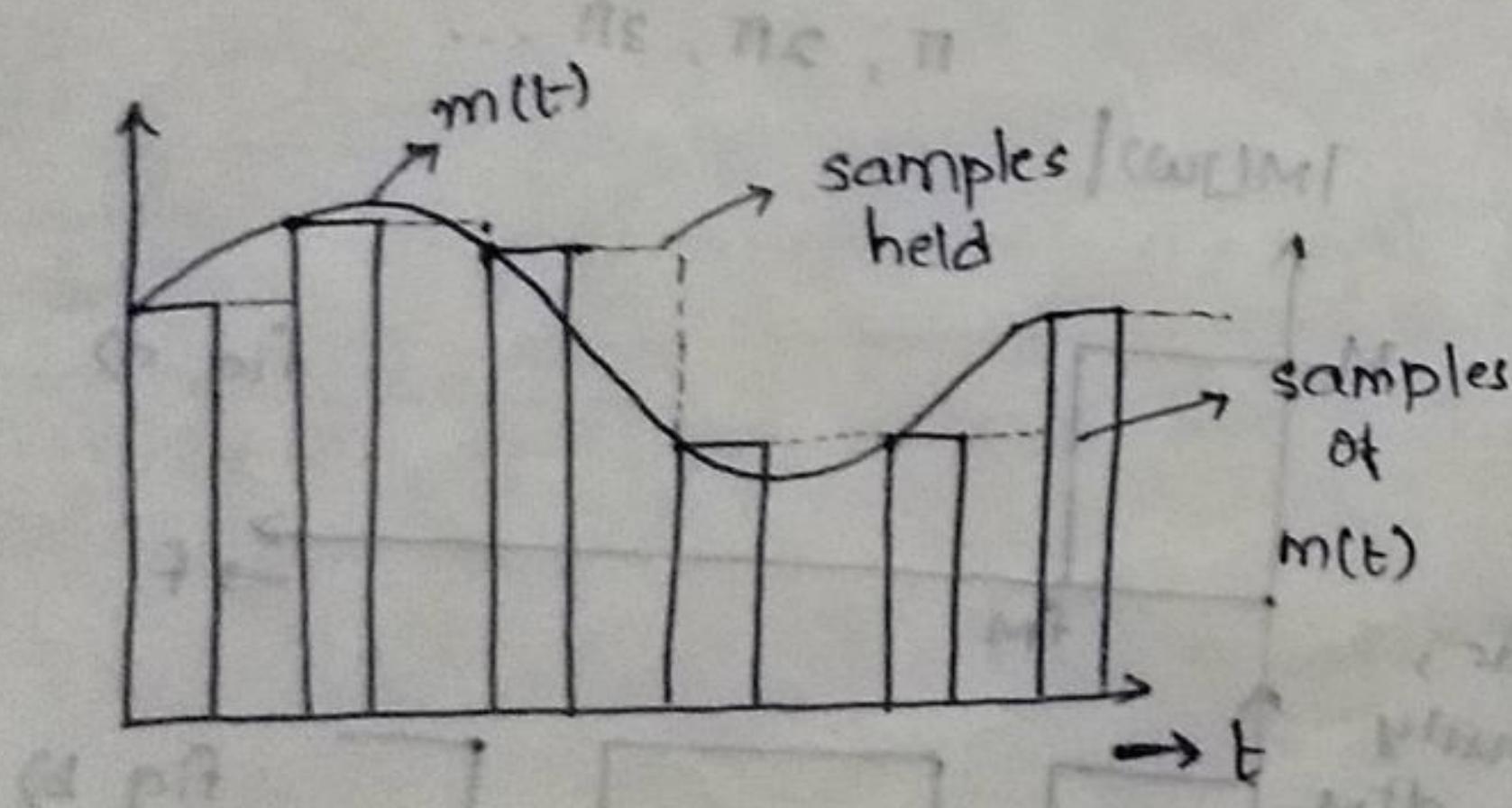
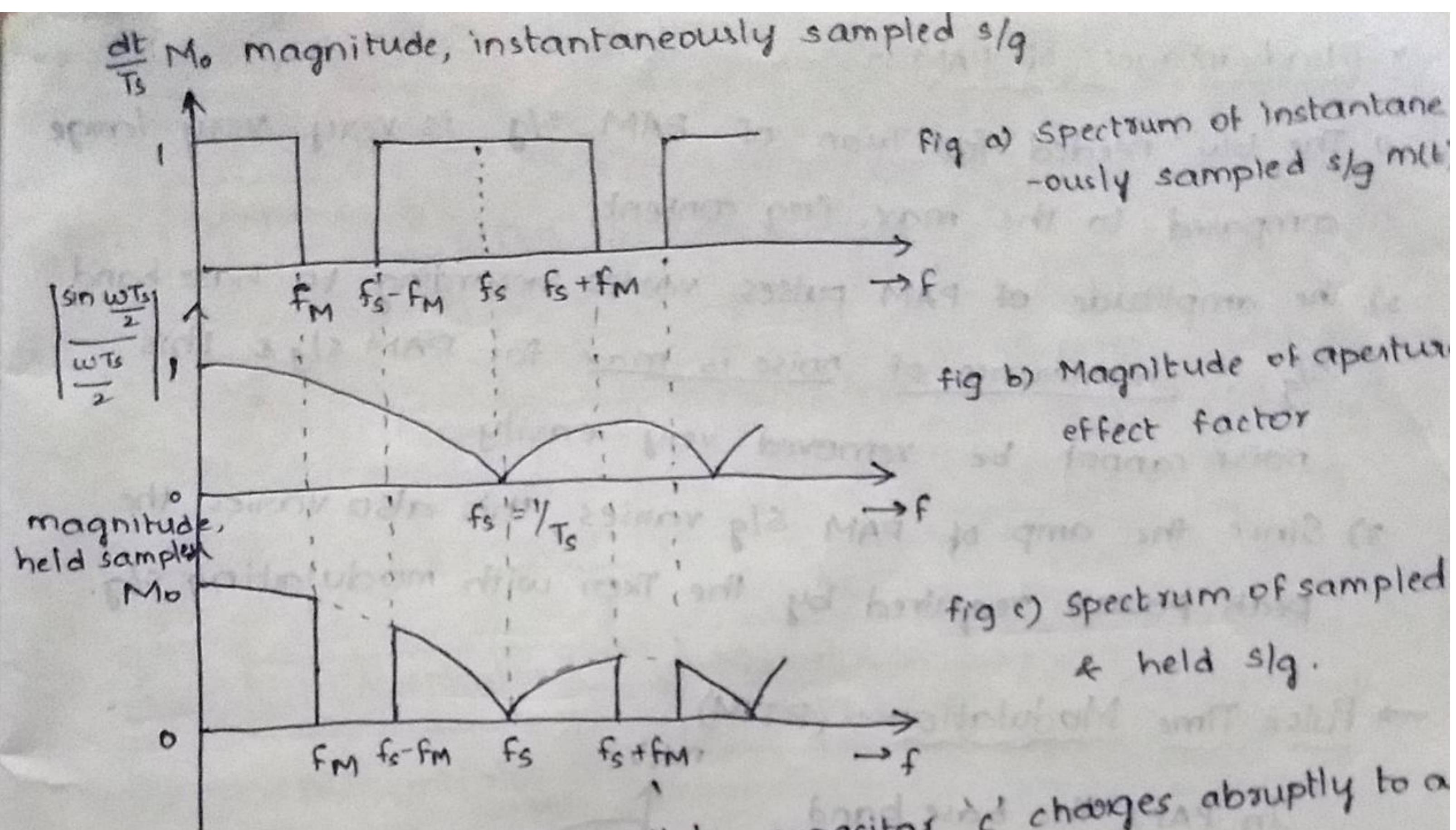


fig : illustrating, operation of holding

At the Rxing end & after demultiplexing, sample value of each individual baseband slg is held until the occurrence of next sample of that same base band slg.

Ckt operation : The switch 'S' operates in synchronism with the occurrence of i/p samples. This switch closes somewhat after the occurrence of leading edge of a sample pulse & opens somewhat before the trailing edge. The amplifier has a low o/p impedance.



At the closing of the switch, capacitor  $C$  charges abruptly to a voltage proportional to the sample value & the capacitor holds this voltage until the operation is repeated for the next sample.

Let the base band s/g  $m(t)$  has spectral density

$$F[m(t)] = M(j\omega)$$

$$F[m(t), \text{sampled \& held}] = \frac{\sin wT_s/2}{wT_s/2} M(j\omega), \text{ or } f \leq f_M$$

flat topped samples stretched to  $T_s$

$$= \frac{T_s}{\tau} F[\text{flat topped samples at } \tau = T_s]$$

The sampled & held s/g has a magnitude larger by a factor  $\frac{T_s}{\tau}$  than the s/g of sample duration  $\tau$ . Thus this method is an alternate method of recovery of base band s/g that raises the level of op s/g.

### Advantages of PAM:

- 1) Easy to implement.
- 2) Greater peak power compared to AM & FM.
- 3) In PAM, pulses are of uniform duration & equally spaced.  
 $\therefore$  Rxer synchronization is not a problem.

## → Disadvantages of PAM :-

- 1) The BW needed for Txion of PAM sig. is very very large compared to the max. freq content.
- 2) The amplitude of PAM pulses varies according to base band sig. ∴ Interference of noise is max for PAM sig & this noise cannot be removed very easily.
- 3) Since the amp. of PAM sig varies, this also varies the peak power required by the Txer with modulating sig.

## → Pulse Time Modulation :- (PTM)

In PAM, the base band

s/g modulates the amplitude of sampling s/g, but in PTM

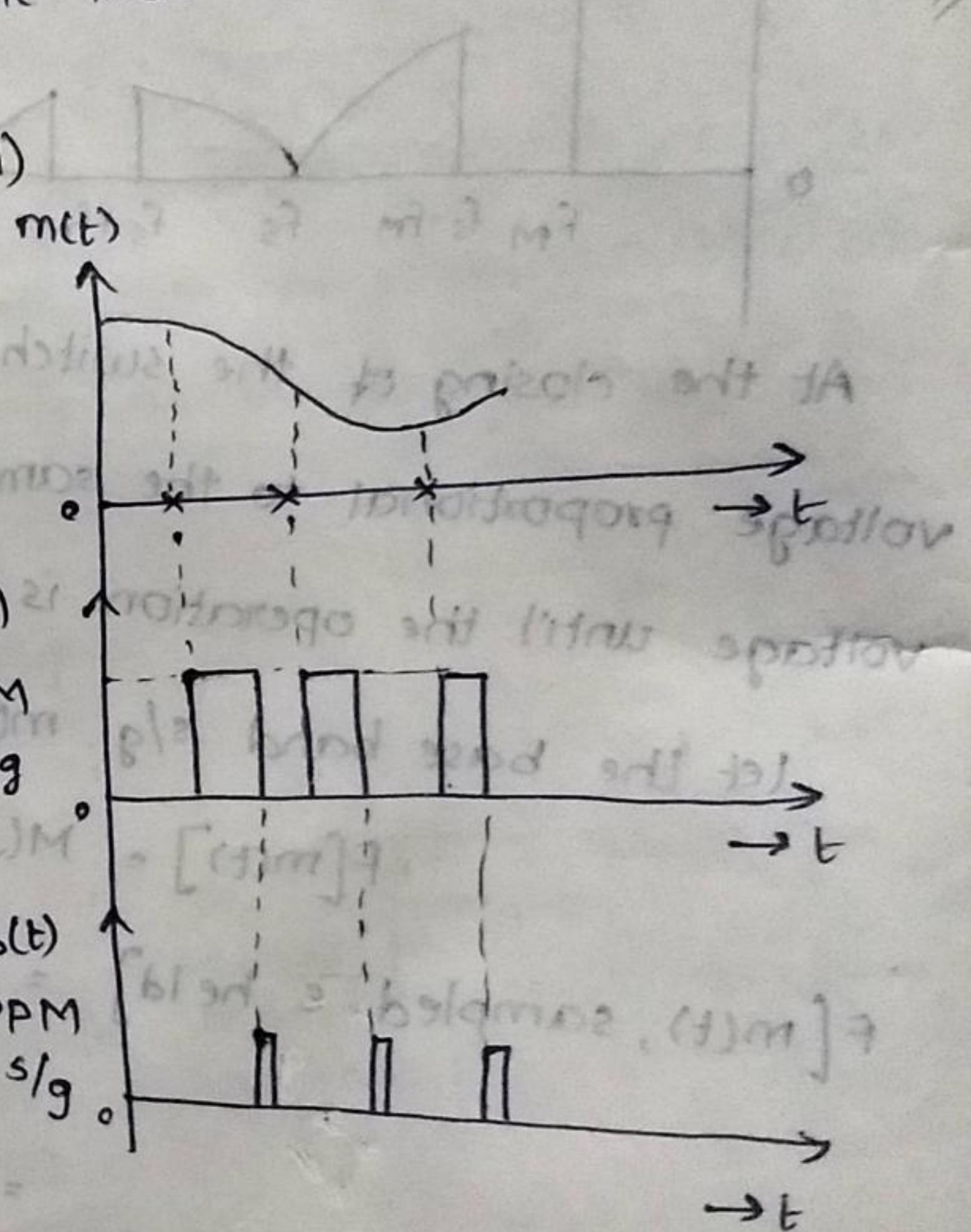
one of the time characteristics (such as width, position) of the sampling s/g is varied

according to the s/g amplitude at that instant.

PTM is of 2 types :-

→ 1) PWM / PDM / PLM

→ 2) PPM



PWM : Pulse width modulation

PDM : Pulse Duration Modulation

PLM : Pulse Length Modulation

In PDM system, the duration of sampled pulse is proportional to sample value of base band s/g amplitude, whereas PPM system generates pulses of constant amplitude & width whose position is varied according to the base band amplitude at each sampling point.