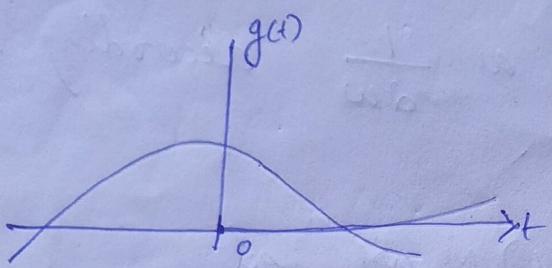
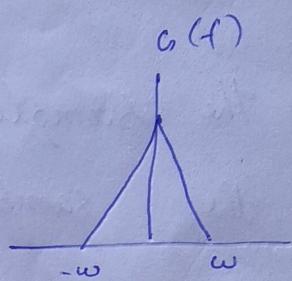


Signal reconstruction & Interpolation formula.

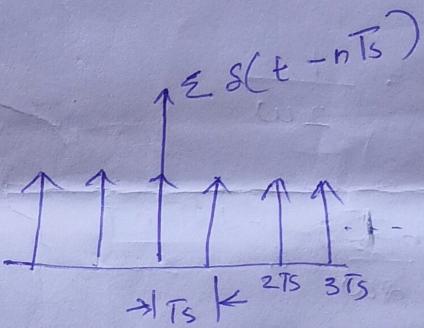
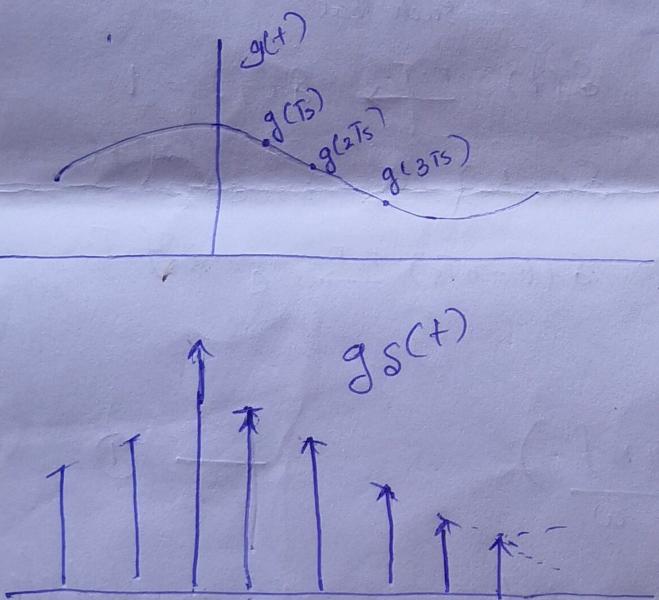
Consider an analog signal $g(t)$ that is continuous in both time and amplitude. This $g(t)$ is band limited to max frequency ω Hz.



and its spectrum



Now we are sampling at an interval of ' T_s ' seconds by multiplying it with a train of impulses $\sum \delta(t - nT_s)$



This is the sampled s/l.

In this sampled s/l the amplitude of train of impulses is getting varied according to the amplitude of $g(t)$ at the sampling instants. The sampled s/l expressed as

$$g_s(t) = \sum_{n=-\infty}^{\infty} g(nT_s) \delta(t - nT_s)$$

In this equation we can say that $g(nT_s)$ is the coefficient of $\delta(t-nT_s)$ or coefficient of expansion which is sequence of amplitude or sequence of numbers.

The sampling interval T_s is taken as $\frac{1}{2\omega}$ according to the sampling theorem

By taking the Fourier transform of $g_s(t)$ we will get-

$$G_s(f) = \sum_{n=-\infty}^{\infty} g(nT_s) e^{-j2\pi n f T_s} \quad \xrightarrow{\text{such that}} \begin{aligned} x(t) &\xrightarrow{\text{F.T.}} X(f) \\ x(t-t_0) &\xrightarrow{\text{F.T.}} X(f) e^{-j2\pi f t_0} \end{aligned}$$

Substitute $T_s = \frac{1}{2\omega}$

$$\begin{aligned} S(t) &\xrightarrow{\text{F.T.}} S(f) = 1 \\ S(t-t_0) &\xrightarrow{\text{F.T.}} 1 \cdot e^{-j2\pi f t_0} \\ \therefore S(t-nT_s) &\xrightarrow{\text{F.T.}} e^{-j2\pi f n T_s} \end{aligned}$$

$$G_s(f) = \sum_{n=-\infty}^{\infty} g\left(\frac{n}{2\omega}\right) \exp\left(-j\frac{\pi}{\omega} n f\right) \quad \text{--- } ①$$

This is the frequency domain representation of the sampled s/l.

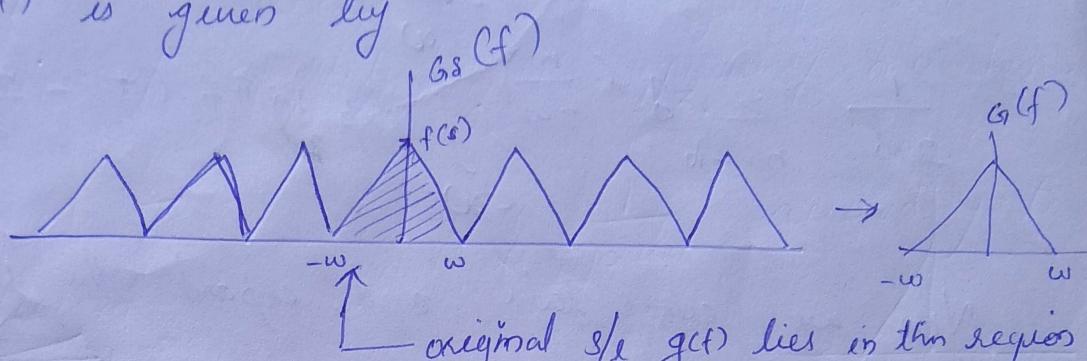
From the sampling theorem we have got another equation

as $G_s(f) = f_s \sum_{n=-\infty}^{\infty} G(f - n f_s) \quad \text{--- } ②$

from the above summation we are taking only zeroth coefficient to the outside of summation

$$G_s(f) = f_s G(f) + \sum_{\substack{n=0 \\ n \neq 0}}^{\infty} G(f - n f_s) \quad \text{--- } ③$$

we have to reconstruct the original signal $g(t)$ from the sampled signal $g_s(t)$; where $g(t)$ is a bandlimited s/e. Spectrum of $g_s(t)$ is given by



original s/e $g(t)$ lies in this region

in order to retain this we have to apply 2 conditions
in eqn no: 3

$$(i) G(f) = 0 \text{ outside the interval } |f| \geq \omega$$

$$(ii) f_s = 2\omega$$

$$G_s(f) = f_s G(f)$$

$$G_s(f) = 2\omega G(f)$$

$$G(f) = \frac{1}{2\omega} G_s(f) \quad \text{--- (4)}$$

Substitute equation no: 1 in eqn no: 4

$$G(f) = \frac{1}{2\omega} \sum_{n=-\infty}^{\infty} g\left(\frac{n}{2\omega}\right) \exp\left(-j\frac{\pi}{\omega}nf\right)$$

$$g(t) = \int_{-\infty}^{\infty} G(f) \exp(j2\pi ft) df.$$

$$= \int_{-\infty}^{\infty} \frac{1}{2\omega} \sum_{n=-\infty}^{\infty} g\left(\frac{n}{2\omega}\right) \exp\left(-j\frac{\pi}{\omega}nf\right) \exp(j2\pi ft) df$$

$$= \frac{1}{2\omega} \sum_{n=-\infty}^{\infty} g\left(\frac{n}{2\omega}\right) \int_{-\infty}^{\infty} \exp\left(-j\frac{\pi}{\omega}nf\right) \exp(j2\pi ft) df$$

$g(nT_s) \rightarrow$ sample values at different time instant

$$= \frac{1}{2\omega} \sum_{n=-\infty}^{\infty} g\left(\frac{n}{2\omega}\right) \int_{-\infty}^{\infty} \exp\left(-j\frac{2\pi n f}{\omega} + j\frac{2\pi f t}{\omega}\right) df.$$

$$= \frac{1}{2\omega} \sum_{n=-\infty}^{\infty} g\left(\frac{n}{2\omega}\right) \int_{-\infty}^{\infty} \exp j\frac{2\pi f}{\omega} \left(t - \frac{n}{2\omega}\right) df \quad \text{--- (5)}$$

Now we can take and expand the integration part only

$$\int_{-\infty}^{\infty} \exp j\frac{2\pi f}{\omega} \left(t - \frac{n}{2\omega}\right) df \rightarrow \text{is the form of } e^{j\theta}$$

$$e^{j\theta} = \cos\theta + j\sin\theta$$

$$\int_{-\infty}^{\infty} \cos j\frac{2\pi f}{\omega} \left(t - \frac{n}{2\omega}\right) + j\sin j\frac{2\pi f}{\omega} \left(t - \frac{n}{2\omega}\right) df$$

$$= 2 \int_0^{\omega} \cos j\frac{2\pi f}{\omega} \left(t - \frac{n}{2\omega}\right) df + 0$$

$$= 2 \left[\frac{\sin j\frac{2\pi f}{\omega} \left(t - \frac{n}{2\omega}\right)}{j\frac{2\pi}{\omega} \left(t - \frac{n}{2\omega}\right)} \right]_0^\omega$$

$\sin 0 = 0$; so we can apply upper limit only.

work that
 $\int_{-\infty}^{\infty} f(\theta) d\theta = 0 \rightarrow$ if the given function is odd.

$\int_{-\infty}^{\infty} f(\theta) d\theta = 2 \int_0^{\infty} f(\theta) d\theta \rightarrow$ if the given function is even.

$\sin\theta$ - odd function

$\cos\theta$ - even function

$$= \frac{\sin j\frac{2\pi \omega}{\omega} \left(t - \frac{n}{2\omega}\right)}{j\frac{2\pi}{\omega} \left(t - \frac{n}{2\omega}\right)} \Rightarrow \frac{2\omega \sin(j2\pi\omega t - n\pi)}{2\pi\omega t - n\pi}$$

Substitute this integration result in eqn no: 5; thus

$$g(t) = \frac{1}{2\omega} \sum_{n=-\infty}^{\infty} g\left(\frac{n}{2\omega}\right) \cdot \frac{2\omega \sin j\frac{2\pi \omega}{\omega} \left(t - \frac{n}{2\omega}\right)}{2\pi\omega t - n\pi}$$

$$g(t) = \sum_{n=-\infty}^{\infty} g\left(\frac{n}{2\pi}\right) \cdot \frac{\sin 2\pi \omega t - n\pi}{2\pi \omega t - n\pi}$$

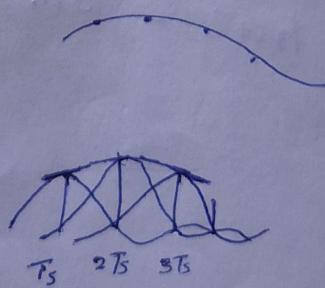
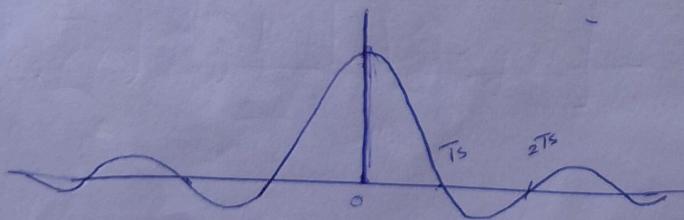
$$\xrightarrow{\quad} \frac{\sin \pi x}{\pi x} = \text{sinc } x$$

$$\therefore g(t) = \sum_{n=-\infty}^{\infty} g\left(\frac{n}{2\pi}\right) \text{sinc}(2\pi \omega t - n)$$

This is known as interpolation formula for reconstructing the original s/e $g(t)$ from the sequence of sample values $g\left(\frac{n}{2\pi}\right)$; with the sinc function playing the role of an interpolation function.

$$\text{sinc } x = \begin{cases} 1 & x=0 \\ 0 & x=\pm 1, \pm 2 \end{cases} \Rightarrow \begin{matrix} \text{interpolatory} \\ \text{property} \end{matrix}$$

Each sample is multiplied by a delayed version of the interpolation function, and all the resulting waveforms are added to obtain $g(t)$.



The process of reconstructing a continuous time s/e $g(t)$ from its samples is called as interpolation. So a signal $g(t)$ band limited to $B_m \text{ Hz}$ or $\omega_m \text{ rad/s}$ can be reconstructed completely from its samples. This is achieved by passing the sampled s/e through an ideal low-pass filter of cut-off frequency $B_m \text{ Hz}$ or $\omega_m \text{ rad/s}$.

gain of the low pass filter will be T_s

where $T_s = \frac{1}{2\omega} - \frac{1}{2f_m} \rightarrow$ [Assuming Sampling is done at Nyquist rate]

Reconstruction and interpolation transfer function may be expressed as

$$H(\omega) = T_s \times \text{rect} \left[\frac{\omega}{4\pi f_m} \right]$$

The impulse response $b(t)$ of this filter is the inverse Fourier transform of $H(\omega)$

$$b(t) = F^{-1}[H(\omega)] = F^{-1} \left[T_s \text{rect} \left(\frac{\omega}{4\pi f_m} \right) \right]$$

$$b(t) = 2 f_m T_s \text{sinc}(2\pi f_m t) \quad \text{--- (A)}$$

$$T_s = \frac{1}{2f_m}$$

$$2 f_m T_s = 1 \rightarrow \text{putting this in eqn A}$$

$$b(t) = \sin c(2\pi f_m t)$$

