



SNS COLLEGE OF TECHNOLOGY
An Autonomous Institution
Coimbatore-35



Accredited by NBA – AICTE and Accredited by NAAC – UGC with 'A++'
Grade

Approved by AICTE, New Delhi & Affiliated to Anna University, Chennai

**DEPARTMENT OF ELECTRONICS & COMMUNICATION
ENGINEERING**

19ECB301-ANALOG AND DIGITAL COMMUNICATION

III YEAR/ V SEMESTER

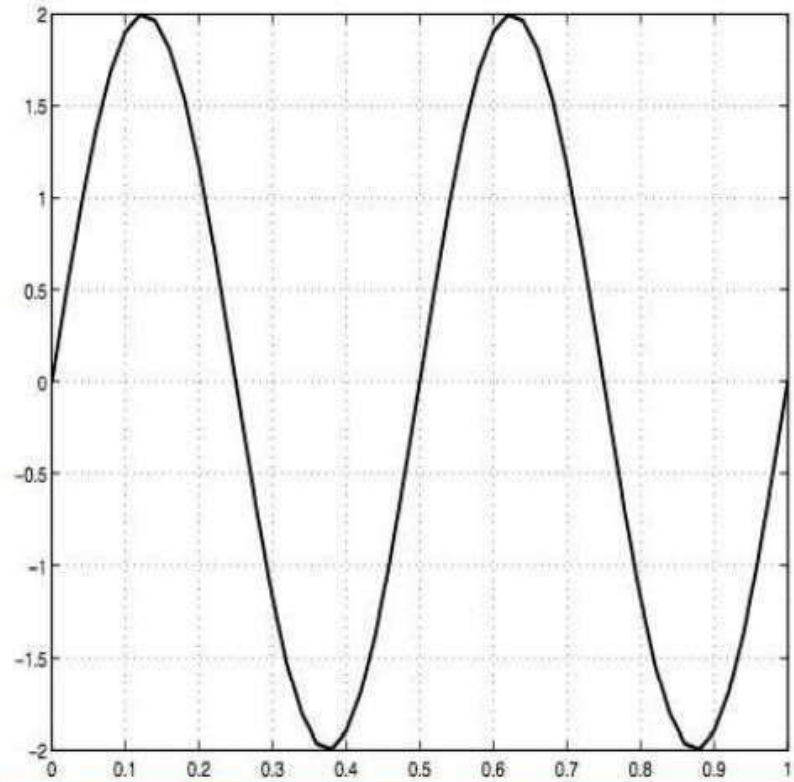
UNIT 3 – DIGITAL COMMUNICATION

TOPIC – SAMPLING

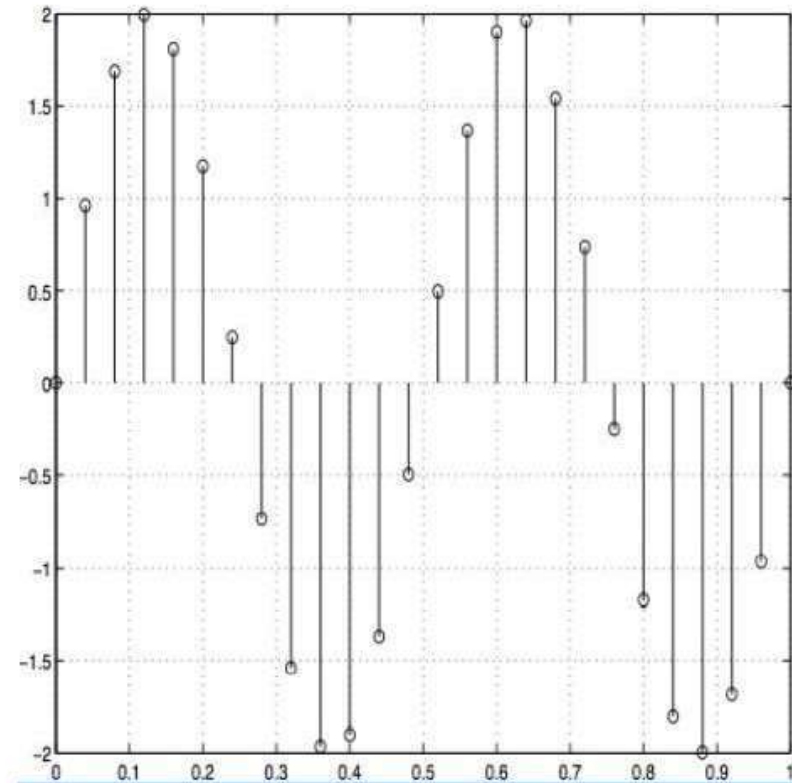
SAMPLING

- Process of converting the continuous time signal to a discrete time signal.
- Sampling is done by taking “Samples” at specific times spaced regularly.
 - $V(t)$ is an analog signal
 - $V(nT_s)$ is the sampled signal
 - T_s = positive real number that represent the spacing of the sampling time
 - n = sample number integer

SAMPLING



Original Analog Signal
"Before Sampling"



Sampled Analog Signal
"After Sampling"

SAMPLING

- The closer the T_s value, the closer the sampled signal resemble the original signal.
- Note that we have lost some values of the original signal, the parts between each successive samples.

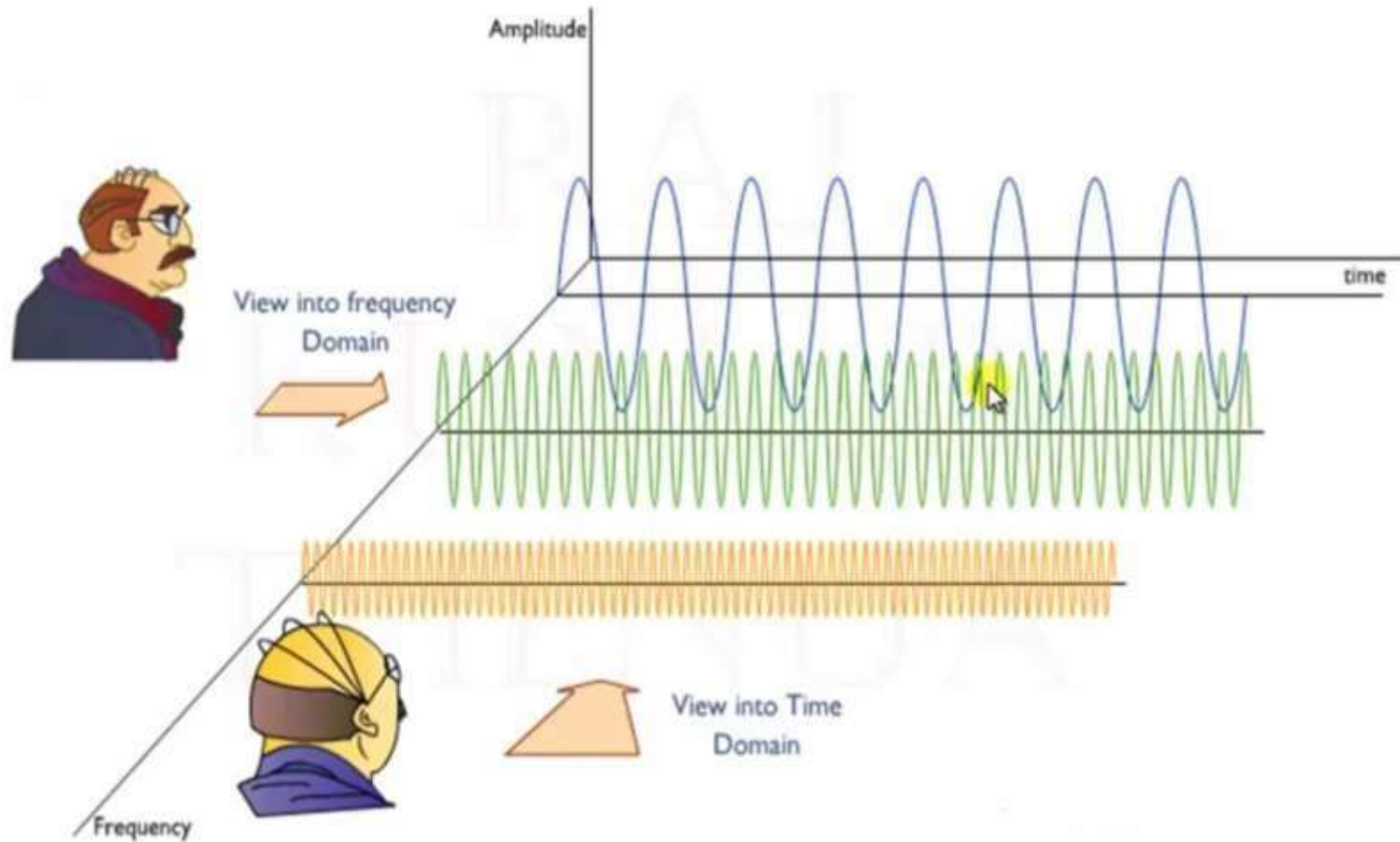
- **Can we recover these values? And How?**
- **Can we go back from the discrete signal to the original continuous signal?**

SAMPLING

- A bandlimited signal having no spectral components above f_{max} (Hz), can be determined uniquely by values sampled at uniform intervals of T_s seconds, where
- An analog signal can be reconstructed from a sampled signal without any loss of information if and only if it is:
 - Band limited signal
 - The sampling frequency is at least twice the signal bandwidth

$$T_s \leq \frac{1}{2f_{max}}$$

TIME DOMAIN Vs FREQUENCY DOMAIN



Sampling theorem - Low pass sampling:-

Consider analog signal $g(t)$
Continuous in time and amplitude.

Infinite duration but finite energy.

Let the sample values of $g(t)$
at times $t = 0, \pm T_s, \pm 2T_s$ denoted by
series as

$$g(nT_s), n = 0, \pm 1, \pm 2, \dots \quad \text{--- (1)}$$

where T_s - Sampling Period

$$f_s = 1/T_s \quad \text{Sampling rate}$$

Discrete time signal $g_s(t)$, results
from the sampling process

$$g_s(t) = \sum_{n=-\infty}^{\infty} g(nT_s) \delta(t - nT_s) \quad \text{--- (2)}$$

$\delta(t - nT_s) \rightarrow$ Dirac delta function
located at $t = nT_s$

Each delta function is weighted
by the corresponding sample value.

from the definition of a delta ¹⁻⁷
function

$$g(nT_s) \delta(t - nT_s) = g(t) \delta(t - nT_s) \text{ --- (3)}$$

Using this, rewrite eqn (2)

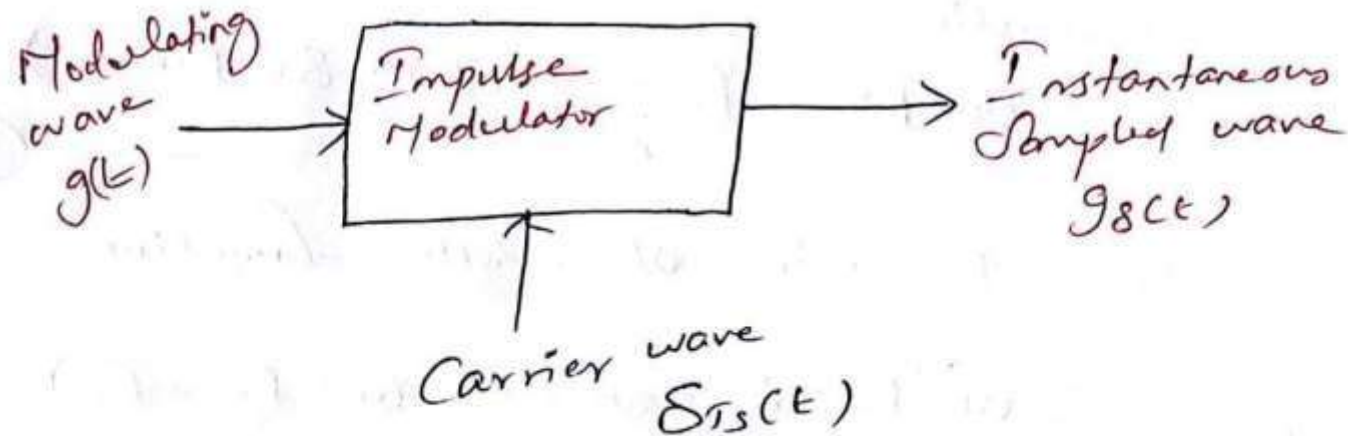
$$g_s(t) = g(t) \sum_{n=-\infty}^{\infty} \delta(t - nT_s)$$

$$g_s(t) = g(t) \delta_{T_s}(t) \text{ --- (4)}$$

$\delta_{T_s}(t)$ = Dirac Combination (or) Ideal
Sampling function.

from eqn (4)

$g_s(t)$ = output of impulse Modulator
operates with $g(t)$ mod. wave
where $\delta_{T_s}(t)$ Carrier wave.



From Fourier theorem

Multiplication of two time function is equivalent to the convolution of their Fourier transform

$$G(f) = \text{FT of } g(t)$$

$$G(g(t)) = \text{FT of } g_s(t)$$

Fourier transform of $\delta_{T_s}(t)$

$$F[\delta_{T_s}(t)] = f_s \sum_{m=-\infty}^{\infty} \delta(f - m f_s) \quad \text{--- (5)}$$

Transforming (4) into frequency domain

$$G_s(f) = G(f) * \left[f_s \sum_{m=-\infty}^{\infty} \delta(f - m f_s) \right]$$

* - Convolution function

Interchanging order of summation & Convolution

$$G_s(f) = f_s \sum_{m=-\infty}^{\infty} G(f) * \delta(f - m f_s) \quad \text{--- (7)}$$

from properties of Delta function

$$G(f) * \delta(f - m f_s) = G(f - m f_s) \quad \text{--- (8)}$$

from eqn (8) simplify eqn (7)

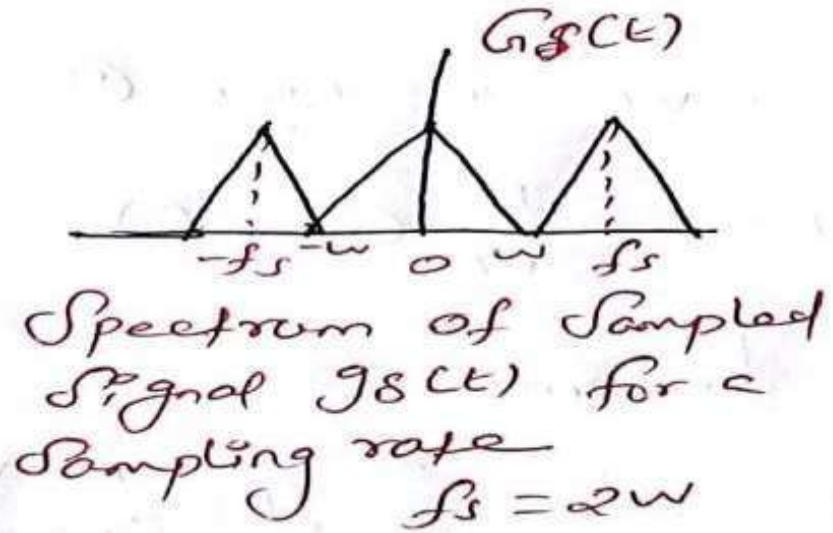
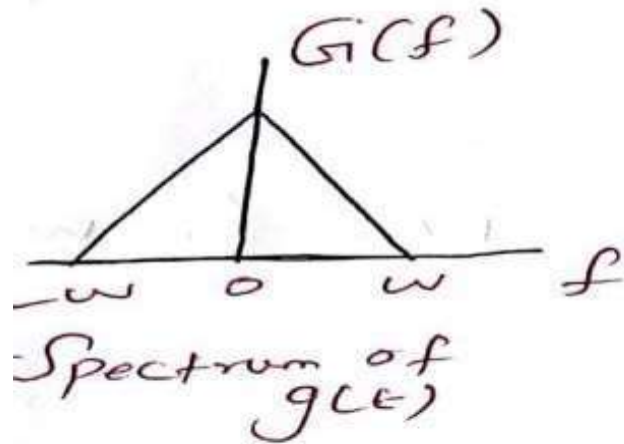
$$G_s(f) = f_s \sum_{m=-\infty}^{\infty} G(f - m f_s) \quad \text{--- (9)}$$

In eqn (9)

$G_s(f)$ - Represents a Spectrum that is periodic in the frequency f with period f_s

- Periodic extension of the Original Spectrum $G(f)$

Process of Uniformly Sampling a signal in the time domain results in a periodic spectrum in the frequency domain with a period equal to the sampling rate.



Another representation obtained by taking Fourier transform on both sides of eqn (2) and

$$F[\delta(t - nT_s)] = \exp[-j2\pi f n T_s]$$

So (2) becomes by applying $F[\delta(t - nT_s)]$

$$G_S(t) = \sum_{n=-\infty}^{\infty} g(nT_s) \exp[-j2\pi f n T_s] \quad \text{--- (10)}$$

when signal has no frequency components higher than ω hertz

then $G(f) = 0$ for $|f| \geq \omega$

Choose Sampling Period $T_s = \frac{1}{2\omega}$

Sub in eqn (10)

$$G_S(f) = \sum_{n=-\infty}^{\infty} g\left(\frac{n}{2\omega}\right) \exp\left(\frac{-j2\pi n f}{2\omega}\right) \quad \text{--- (11)}$$

Sub $f_s = 2W$ in eqn (9)

$$G(f) = \frac{1}{2W} G_8(f) \quad -W < f < W$$

So (10) becomes

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

(12)

THANK YOU