

## Sampling Theorem and its Importance

- Sampling Theorem:

*“A bandlimited signal can be reconstructed exactly if it is sampled at a rate atleast twice the maximum frequency component in it.”*

Figure 1 shows a signal  $g(t)$  that is bandlimited.

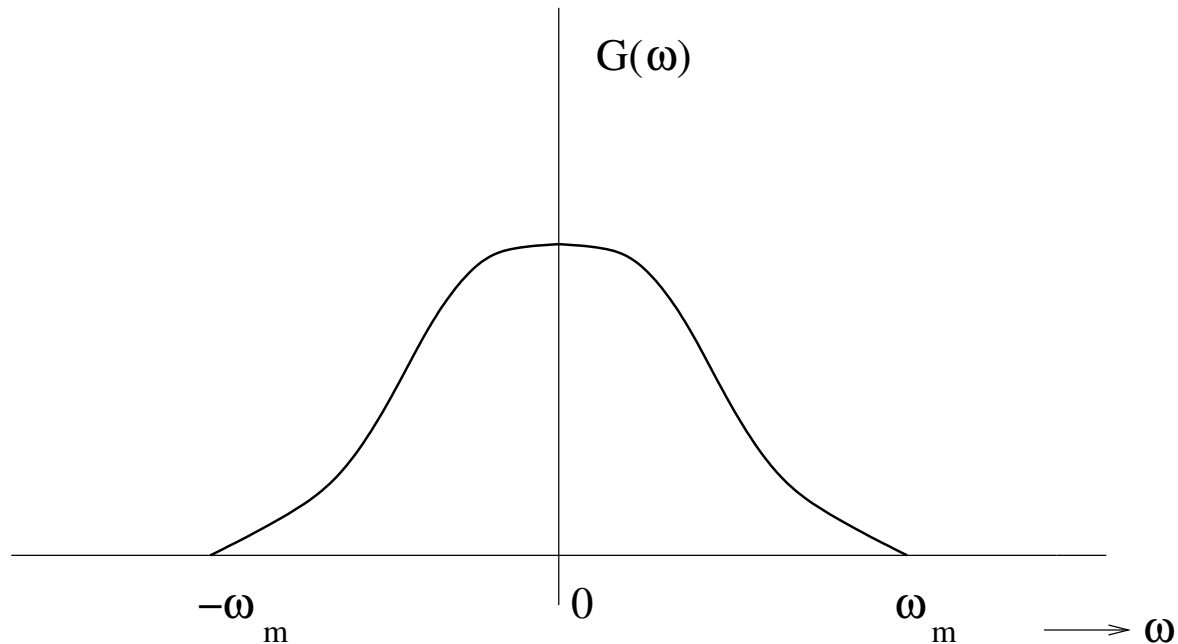


Figure 1: Spectrum of bandlimited signal  $g(t)$

- The maximum frequency component of  $g(t)$  is  $f_m$ . To recover the signal  $g(t)$  exactly from its samples it has to be sampled at a rate  $f_s \geq 2f_m$ .
- The minimum required sampling rate  $f_s = 2f_m$  is called

*Nyquist rate.*

**Proof:** Let  $g(t)$  be a bandlimited signal whose bandwidth is  $f_m$  ( $\omega_m = 2\pi f_m$ ).

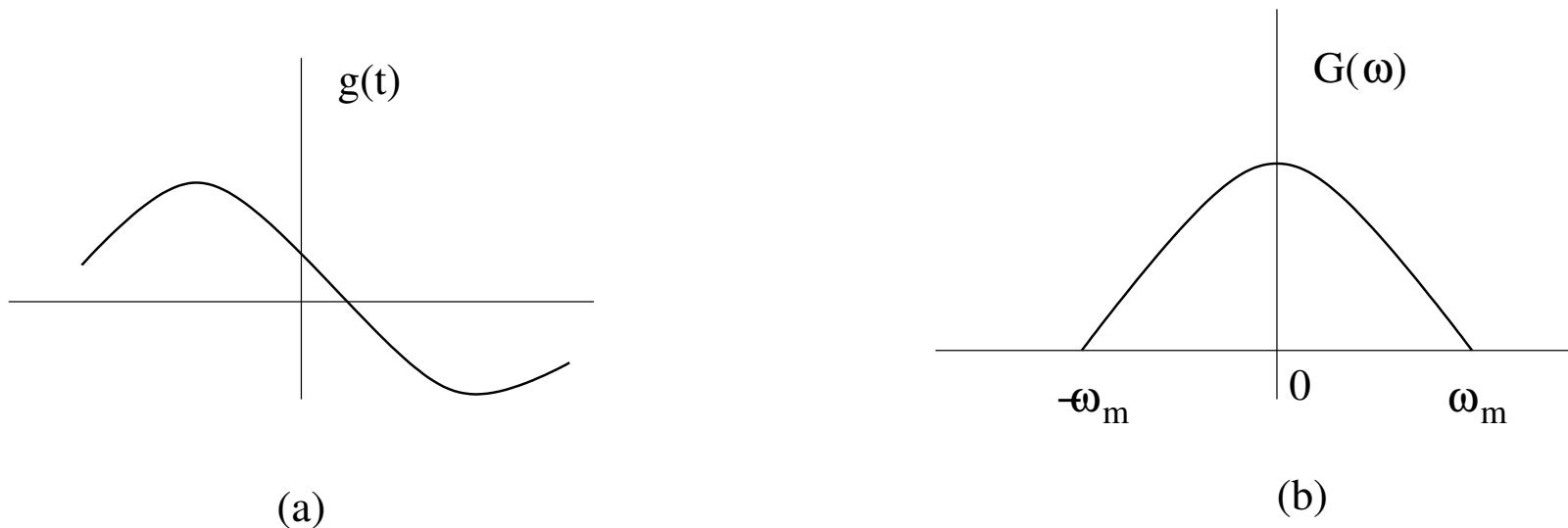


Figure 2: (a) Original signal  $g(t)$  (b) Spectrum  $G(\omega)$

$\delta_T(t)$  is the sampling signal with  $f_s = 1/T > 2f_m$ .

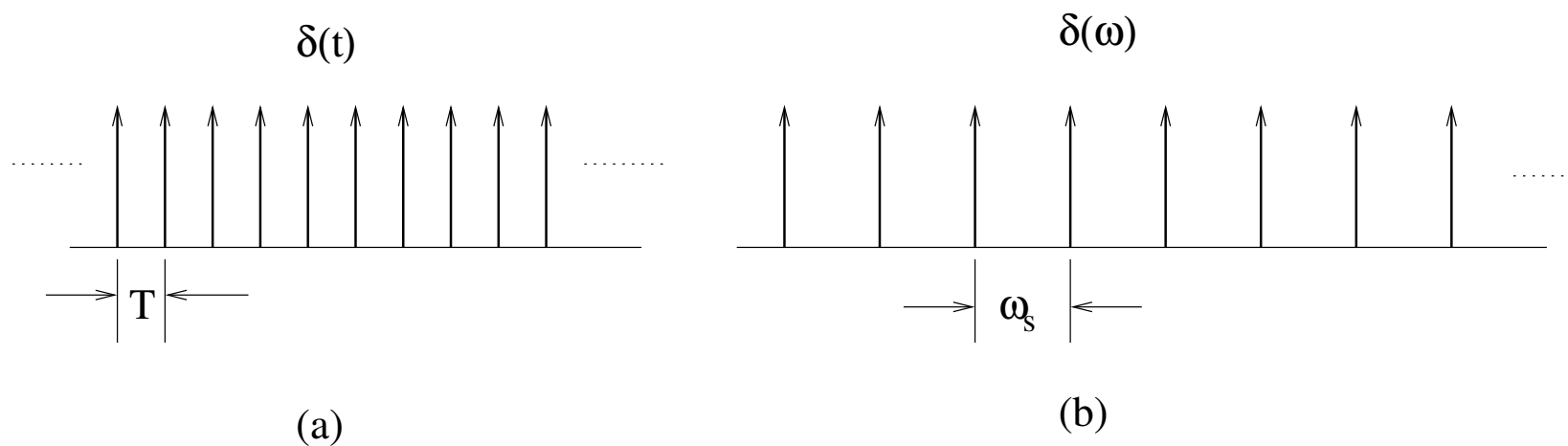


Figure 3: (a) sampling signal  $\delta_T(t)$  (b) Spectrum  $\delta_T(\omega)$

- Let  $g_s(t)$  be the sampled signal. Its Fourier Transform  $G_s(\omega)$  is given by

$$\begin{aligned}
\mathcal{F}(g_s(t)) &= \mathcal{F}[g(t)\delta_T(t)] \\
&= \mathcal{F}\left[g(t) \sum_{n=-\infty}^{+\infty} \delta(t - nT)\right] \\
&= \frac{1}{2\pi} \left[ G(\omega) * \omega_0 \sum_{n=-\infty}^{+\infty} \delta(\omega - n\omega_0) \right]
\end{aligned}$$

$$G_s(\omega) = \frac{1}{T} \sum_{n=-\infty}^{+\infty} G(\omega) * \delta(\omega - n\omega_0)$$

$$G_s(\omega) = \mathcal{F}[g(t) + 2g(t) \cos(\omega_0 t) + 2g(t) \cos(2\omega_0 t) + \cdots]$$

$$G_s(\omega) = \frac{1}{T} \sum_{n=-\infty}^{+\infty} G(\omega - n\omega_0)$$

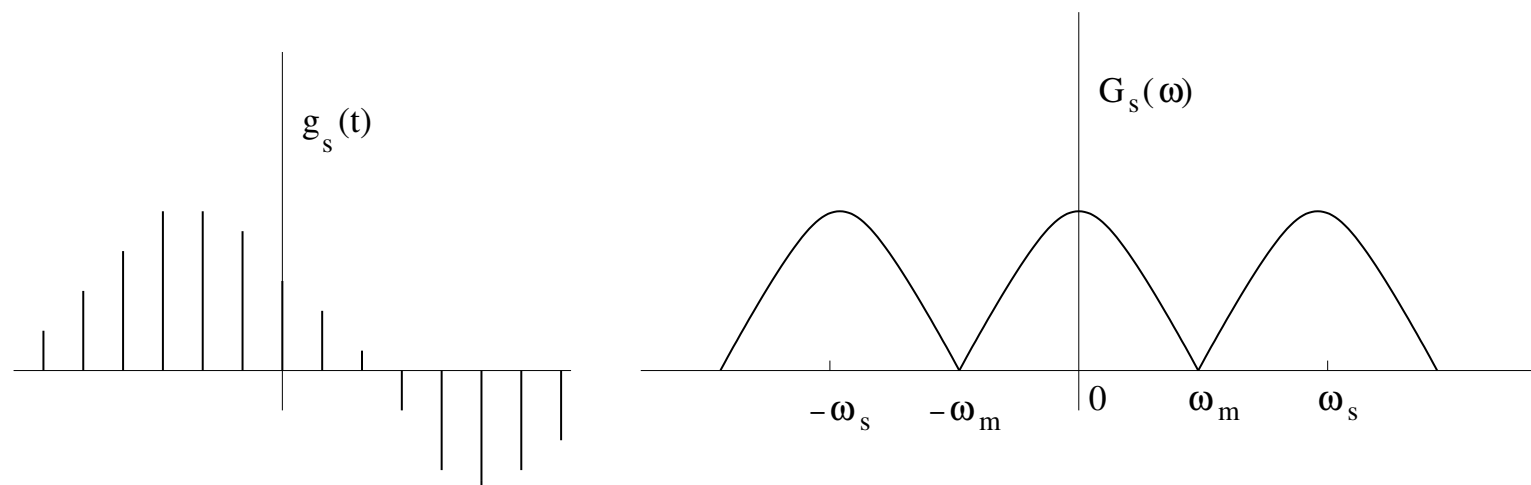


Figure 4: (a) sampled signal  $g_s(t)$  (b) Spectrum  $G_s(\omega)$

- If  $\omega_s = 2\omega_m$ , i.e.,  $T = 1/2f_m$ . Therefore,  $G_s(\omega)$  is given by

$$G_s(\omega) = \frac{1}{T} \sum_{n=-\infty}^{+\infty} G(\omega - n\omega_m)$$

- To recover the original signal  $G(\omega)$ :
  1. Filter with a Gate function,  $H_{2\omega_m}(\omega)$  of width  $2\omega_m$ .

2. Scale it by  $T$ .

$$G(\omega) = TG_s(\omega)H_{2\omega_m}(\omega).$$

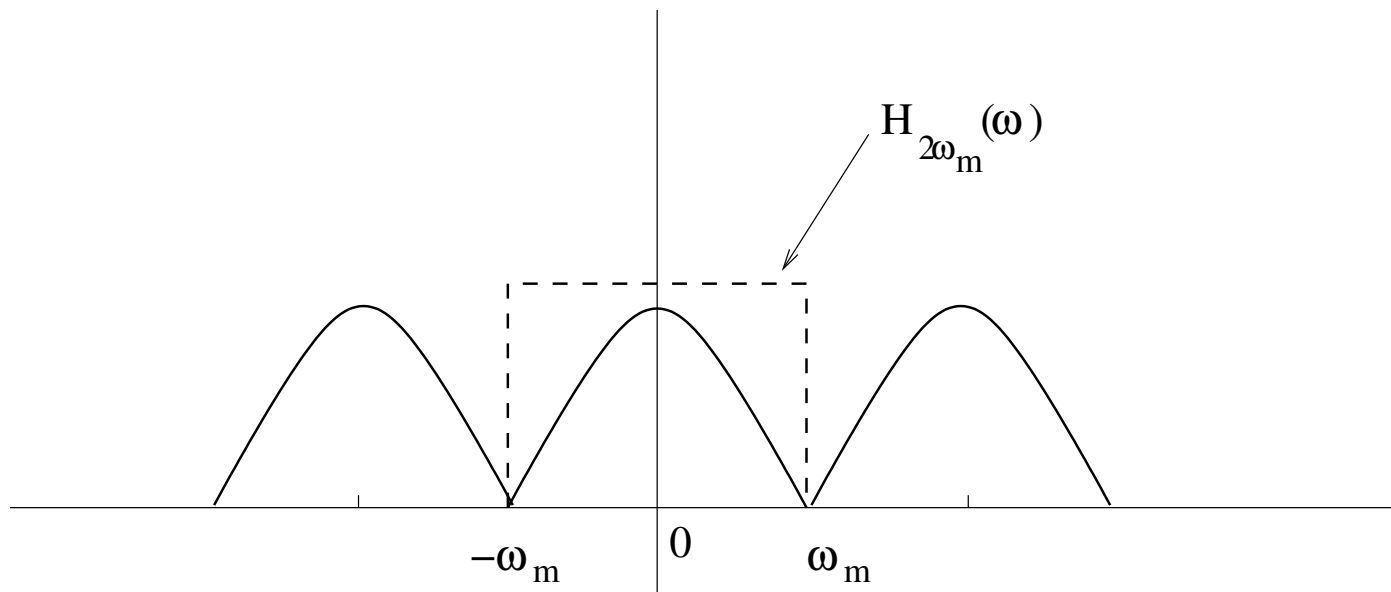


Figure 5: Recovery of signal by filtering with a filter of width  $2\omega_m$

- Aliasing
  - Aliasing is a phenomenon where the high frequency components of the sampled signal interfere with each other

because of inadequate sampling  $\omega_s < 2\omega_m$ .

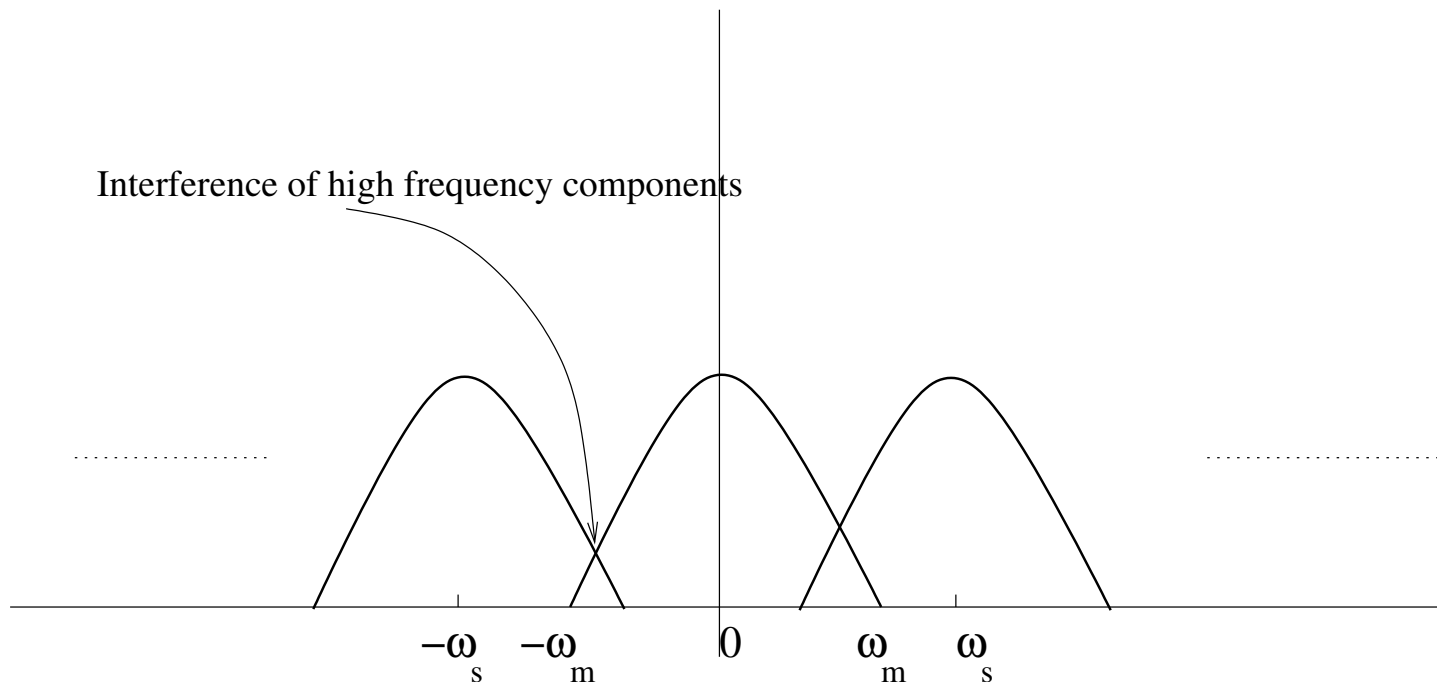


Figure 6: Aliasing due to inadequate sampling

Aliasing leads to distortion in recovered signal. This is the reason why sampling frequency should be at least twice the bandwidth of the signal.



- Oversampling

- In practice signal are oversampled, where  $f_s$  is significantly higher than Nyquist rate to avoid aliasing.

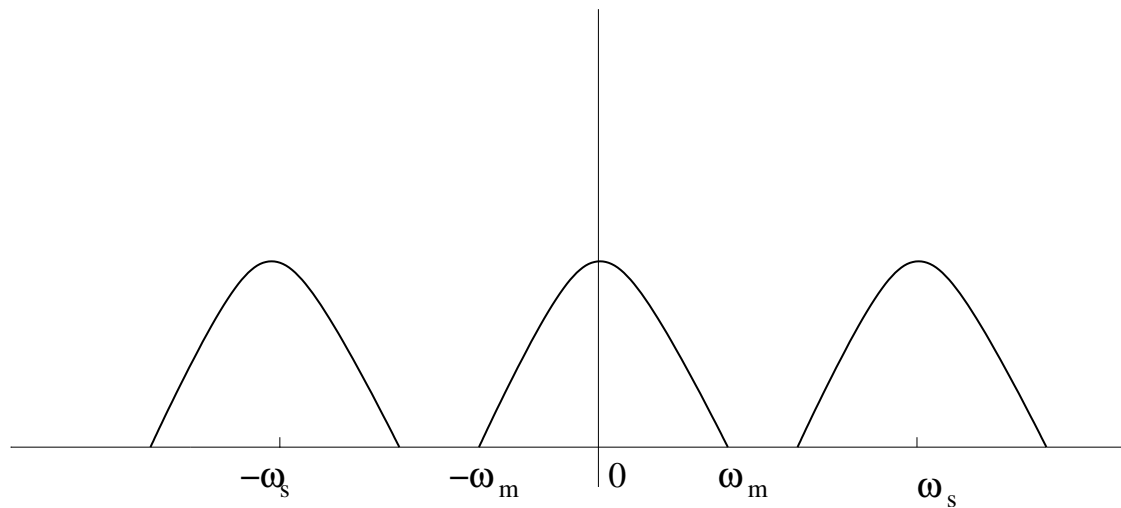


Figure 7: Oversampled signal-avoids aliasing

Problem: Define the frequency domain equivalent of the *Sampling Theorem* and prove it.