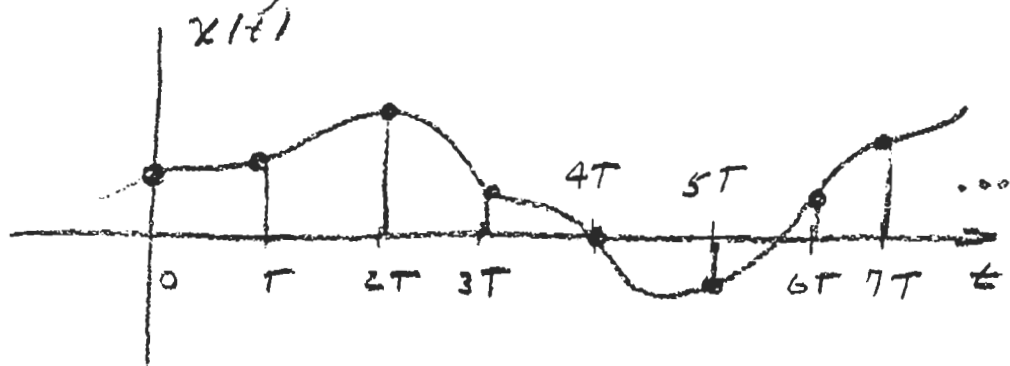


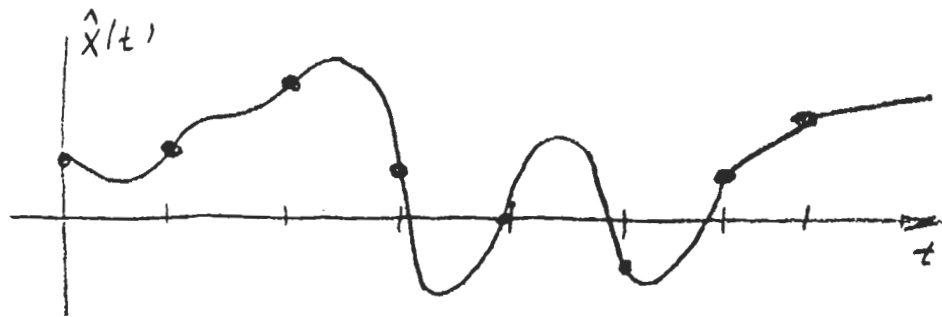
LECTURE S24

Sampling

Especially when working with computers, it is useful to represent a signal by sampling it:



Problem — the samples do not uniquely determine the signal:

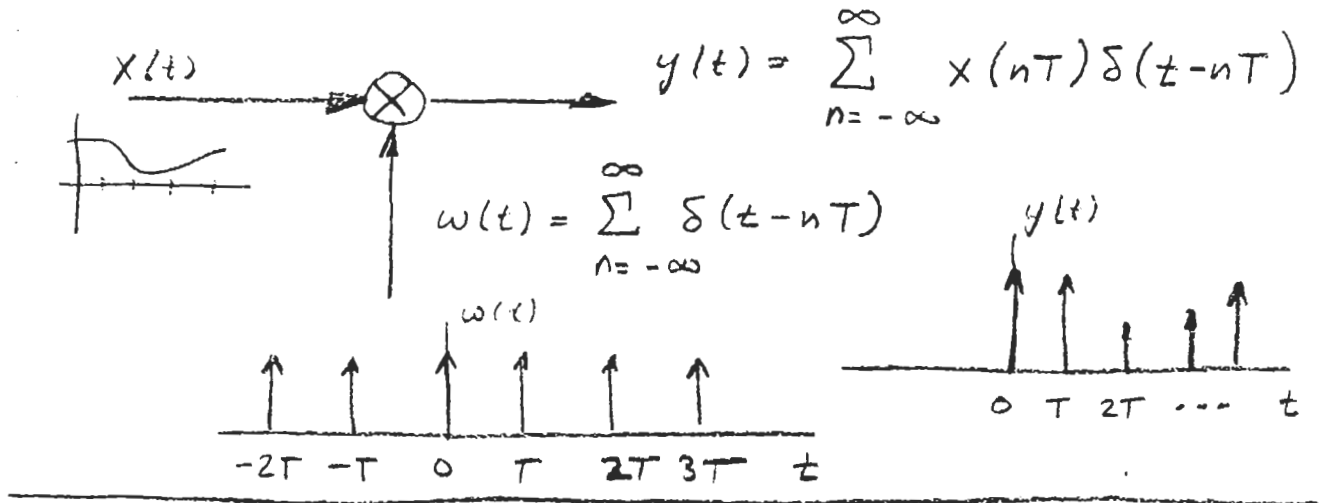


What is the right way to reconstruct the original signal?

How rapidly should we sample?

Modelling the Sampling Process

We can model the sampling process as a kind of modulation:



The FT of $y(t)$ is

$$Y(f) = X(f) * W(f)$$

What is $W(f) = \mathcal{F}[w(t)]$? Answer is a little tricky.

Because $w(t)$ is periodic, can represent by a Fourier Series:

$$w(t) = \sum_{n=-\infty}^{\infty} a_n e^{j2n\pi t/T}$$

where $a_n = \frac{1}{T} \int_{-T/2}^{T/2} w(t) e^{-j2n\pi t/T} dt$

$$a_n = \frac{1}{T} \int_{-T/2}^{T/2} \delta(t) e^{-j2n\pi t/T} dt$$

$$= \frac{1}{T}, \text{ for all } n.$$

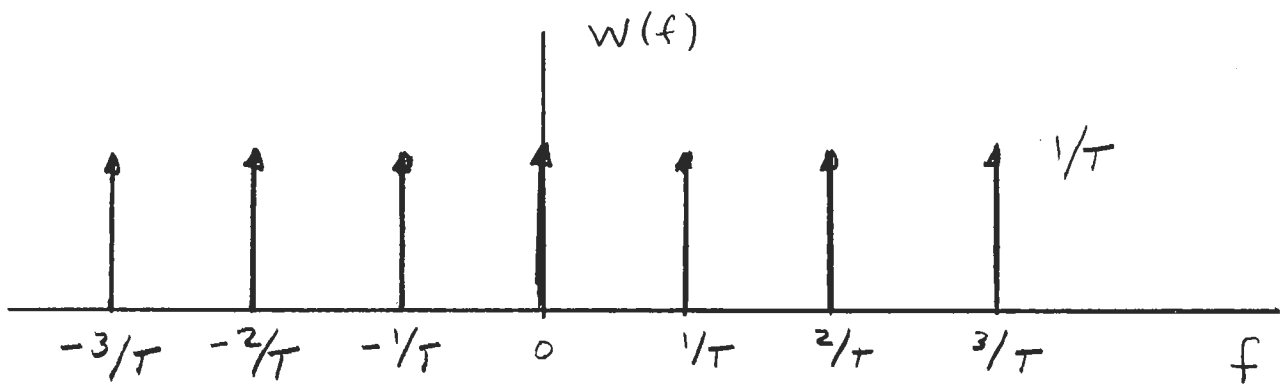
$$\therefore, w(t) = \frac{1}{T} \sum_{n=-\infty}^{\infty} e^{j2n\pi t/T}$$

Note: The sum above doesn't converge
Let's ignore this.

Then $W(f) = \mathcal{F}[w(t)]$

$$= \mathcal{F}\left[\frac{1}{T} \sum_{n=-\infty}^{\infty} e^{j2\pi n t/T}\right]$$

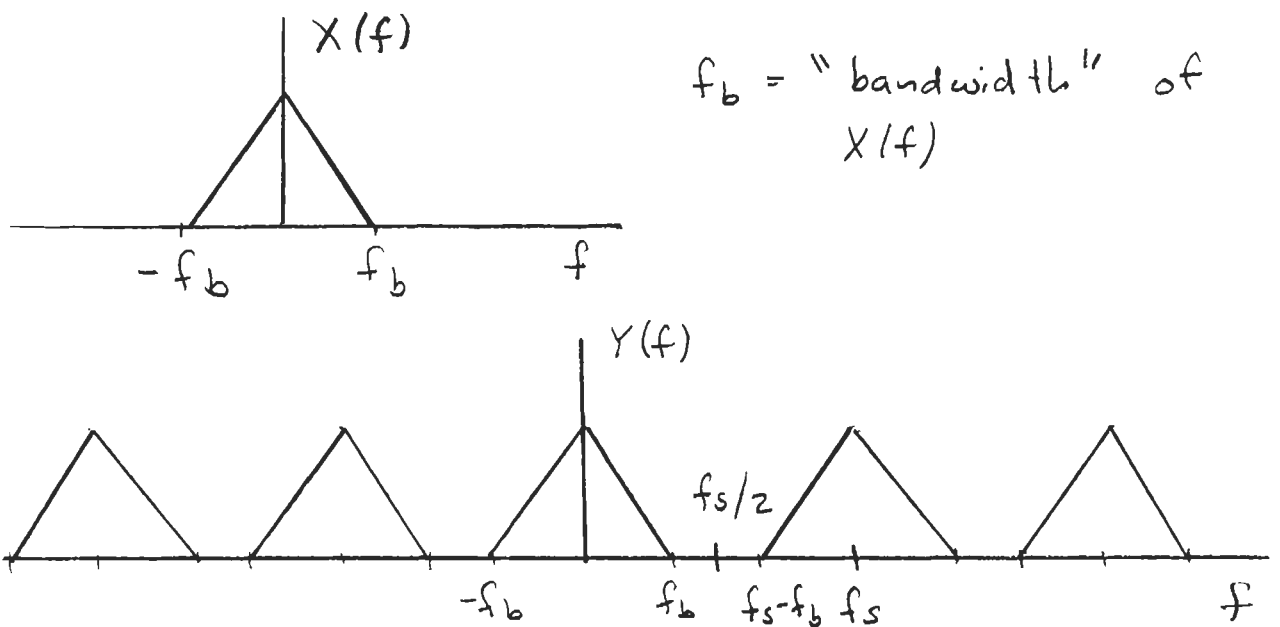
$$= \frac{1}{T} \sum_{n=-\infty}^{\infty} \delta(f - n/T)$$



The FT of a periodic pulse train is a periodic pulse train!

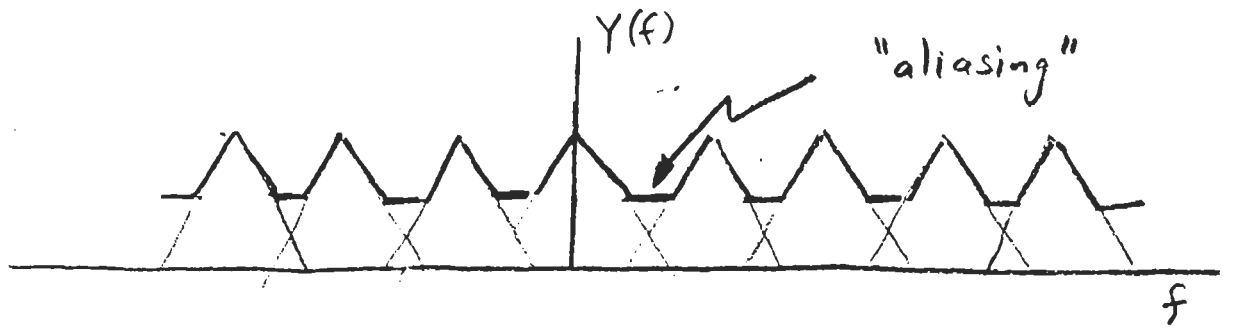
So,

$$\begin{aligned} Y(f) &= X(f) * W(f) \\ &= X(f) * \frac{1}{T} \sum_{n=-\infty}^{\infty} \delta(f - n/T) \\ &= \frac{1}{T} \sum_{n=-\infty}^{\infty} X(f - n/T) \end{aligned}$$



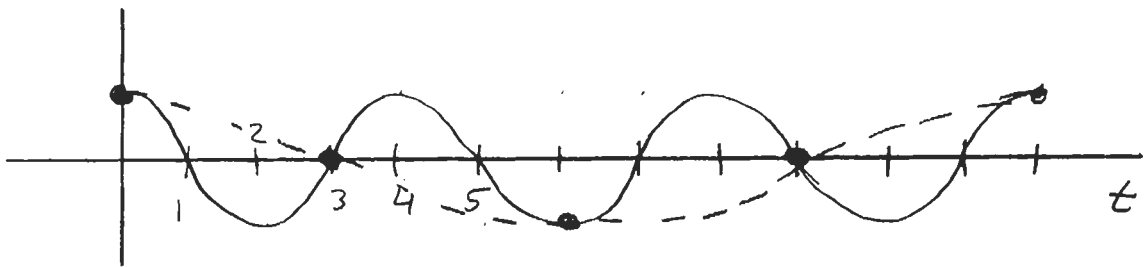
So if $f_s > 2f_b$, we have (many!) exact copies of $X(f)$.

What happens if $f_s < 2f_b$?

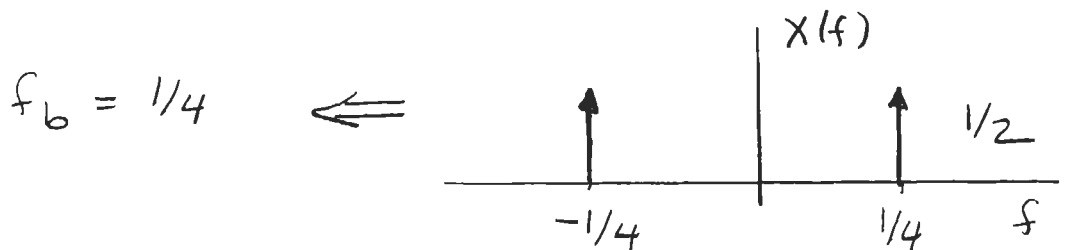


Spectrum of $X(f)$ is no longer intact, so information is lost.

Example



$$x(t) = \cos \frac{\pi t}{2} \quad X(f) = \frac{1}{2} \left(\delta(f - 1/4) + \delta(f + 1/4) \right)$$



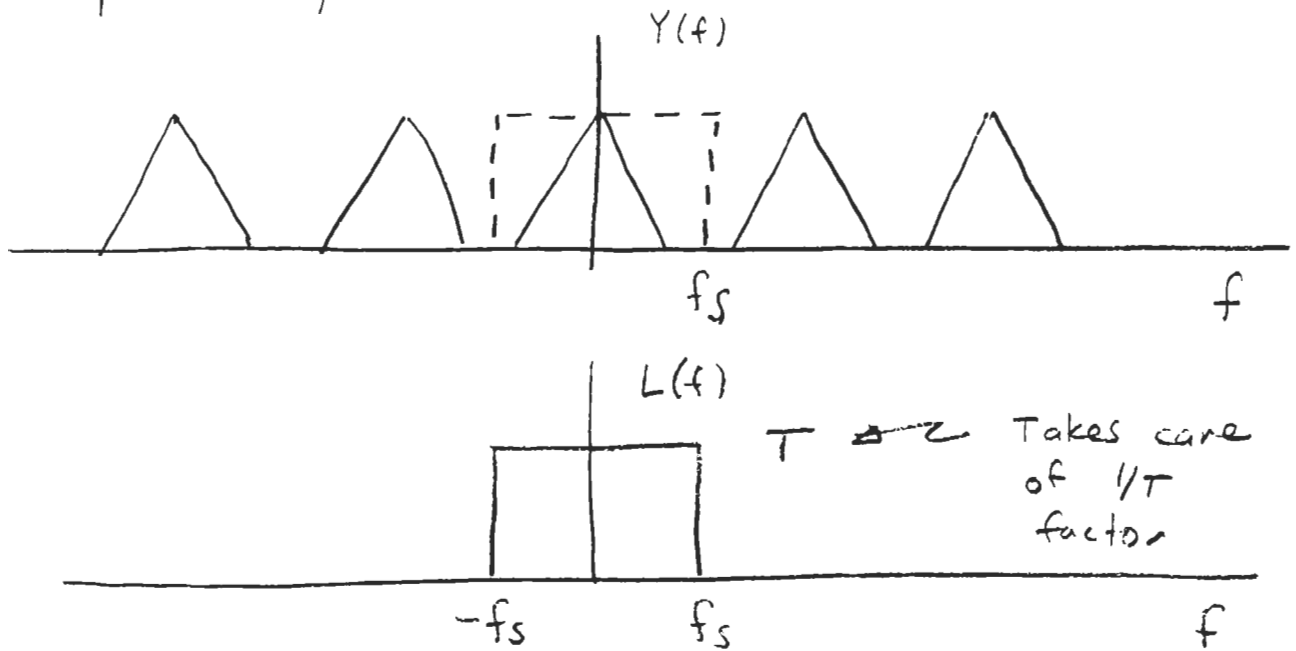
If $T=3$, $f_s = 1/3 < 2 \cdot f_b$. So aliasing occurs. Samples signal looks like it comes from

$$\cos \frac{\pi}{6} t = \text{"alias"} \text{ of } \cos \frac{\pi t}{2}$$

$(f = 1/12) \qquad \qquad \qquad (f = 1/4)$

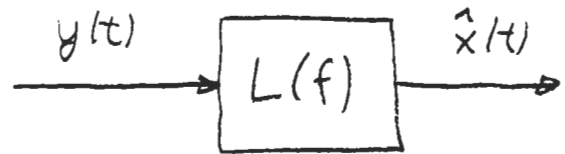
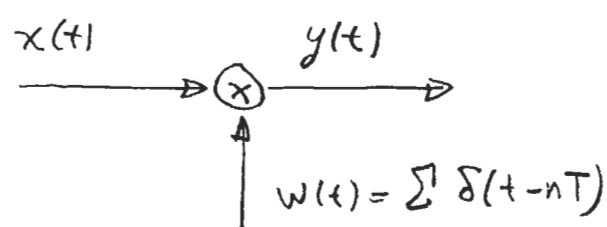
So assume $f_s > 2 f_b =$ "Nyquist frequency"

How can we reconstruct the signal?
Use a low-pass filter to eliminate repeated spectra.



$$L(f) = \begin{cases} T, & |f| \leq f_s/2 \\ 0, & |f| \geq f_s/2 \end{cases}$$

So we have:



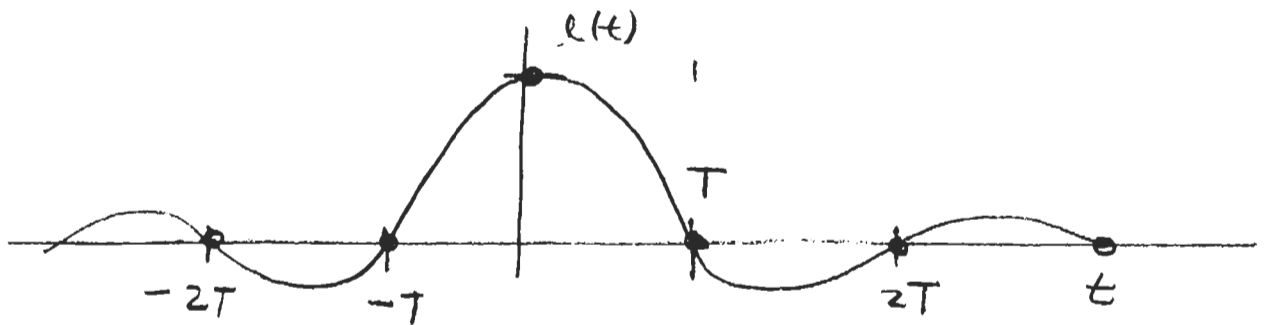
Sampling

Reconstruction

Impulse response of filter is

$$h(t) = \mathcal{F}^{-1}[L(f)]$$

$$= \frac{\text{sinc } \pi t/T}{\pi t/T} = \text{sinc } \frac{\pi t}{T}$$



So reconstructed signal is

$$\hat{x}(t) = \sum_{n=-\infty}^{\infty} x(nT) \text{sinc}\left(\frac{\pi}{T}(t-nT)\right)$$