

## Reconstruction of signal

### The Ideal Low-Pass Filter:

In this lecture, we examine the Ideal low pass filter and the process of reconstruction of a Band-limited signal.

Let us first see another way of interpreting the action of the Ideal low-pass filter.

### Impulse response of Ideal low pass filter:

The Frequency response (the Fourier transform of the impulse response of an LSI system is also called its frequency response) of an ideal low pass filter which allows a bandwidth  $B$ , is a rectangle extending from  $-B$  to  $+B$ , having a constant height as shown in the figure.

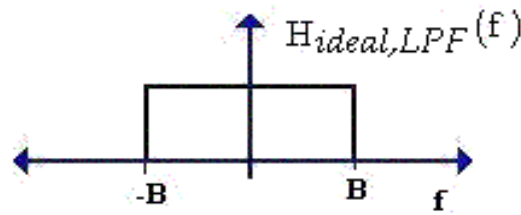


Figure 5.3

Lets look at the Impulse Response of this Ideal low pass filter, taking its height in  $[-B, B]$  to be 1. Using the formula for inverse Fourier Transform we have :

$$\begin{aligned}
 \text{Impulse Response} &= \int_{-B}^B e^{j2\pi ft} df \\
 &= \frac{2B \sin \pi (2Bt)}{\pi (2Bt)} \\
 &= 2B \text{Sinc}(2Bt)
 \end{aligned}$$

$$\text{(note that } \text{sinc}(t) = \frac{\sin(\pi t)}{\pi t} \text{ )}$$

Thus the impulse response of an ideal low pass filter turns out to be a **Sinc function**, which looks like:

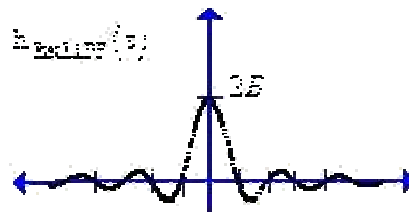


Figure 5.4

### Reconstruction of a signal by low pass filter :

Consider a signal  $x(t)$  having bandwidth less than  $B$ .

We sample  $x(t)$  at a rate  $2B$  and pass  $\tilde{x}(t) = \sum_{n=-\infty}^{\infty} \frac{1}{2B} x\left(\frac{n}{2B}\right) \delta\left(t - \frac{n}{2B}\right)$  into an Ideal low-pass filter of bandwidth  $B$ . Assume  $T_0 = \frac{1}{2B}$ .

The signal  $x(t)$  and the signal  $\tilde{x}(t)$  obtained by multiplying the signal by a periodic train of impulses, separated by  $T_0$ , having strength 1 are shown below.

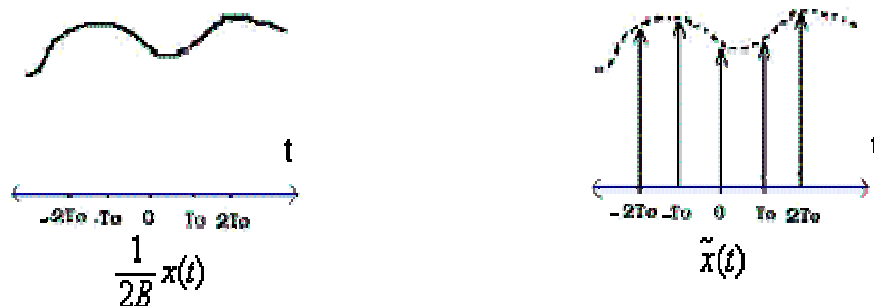


Figure 5.5

What happens when  $\tilde{x}(t)$  is fed into the LSI system?

Lets look at the convolution of the impulse response  $\mathbf{h(t)}$  of the Ideal low-pass filter with  $\tilde{x}(t)$

$$\begin{aligned}\tilde{x}(t) * h(t) &= \left\{ \sum_{n=-\infty}^{\infty} T_0 x(nT_0) \delta(t - nT_0) \right\} * h(t) \\ &= \sum_{n=-\infty}^{\infty} T_0 x(nT_0) h(t - nT_0)\end{aligned}$$

where we have seen  $h(t) = 2B \operatorname{sinc}(2Bt) = \frac{1}{T_0} \operatorname{sinc}\left(\frac{t}{T_0}\right)$

$$\therefore \tilde{x}(t) * h(t) = \sum_{n=-\infty}^{\infty} x(nT_0) \operatorname{sinc}\left(\frac{t}{T_0} - n\right)$$

When  $\tilde{x}(t)$  is passed through a Low pass filter, the output which is the reconstructed signal is nothing but the sum of copies of the impulse response  $\mathbf{h(t)}$  shifted by integral multiples of  $T_0$  and multiplied by the value of  $\mathbf{x(t)}$  at the corresponding integral multiple of  $T_0$ . Also observe that the  $\mathbf{h(t)}$  is zero at all sample points (which are integral multiples of  $T_0$ ) except at zero. Thus, the reconstruction of  $\mathbf{x(t)}$  can be visualized as a sum of the following signals :

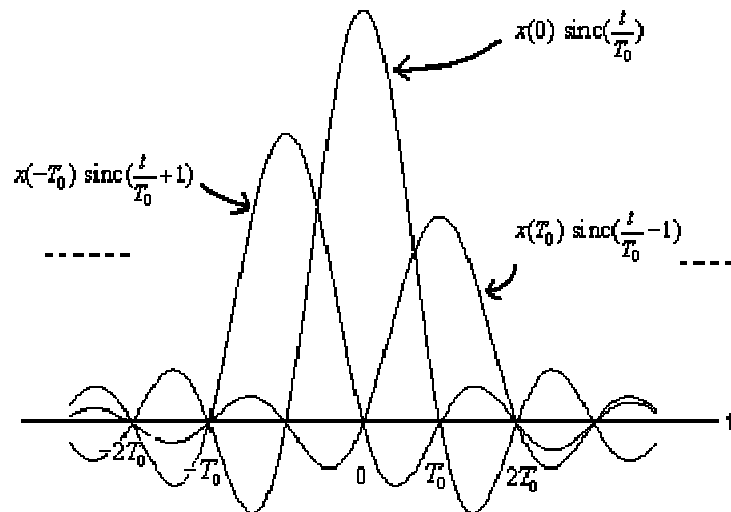


Figure 5.6