

Signal Analysis & Communication ECE 355

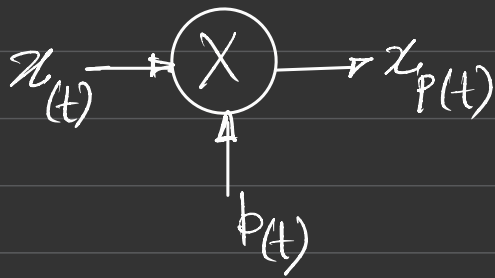
Ch. 7.2: RECONSTRUCTION FROM SAMPLES

Lecture 29

22-11-2023

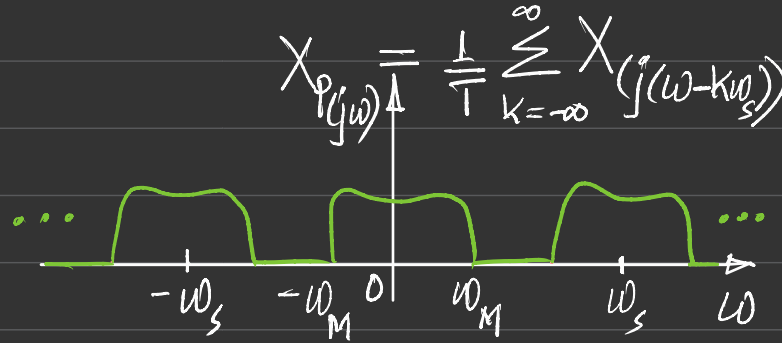
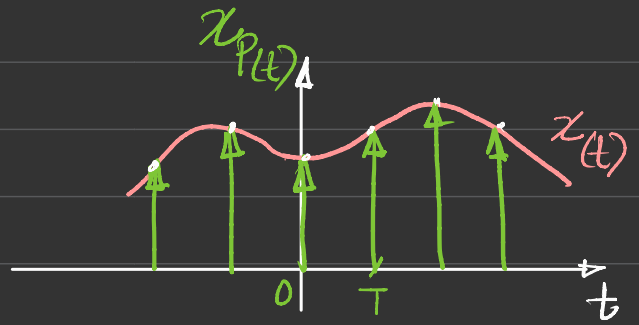


Recall: Impulse Train Sampling

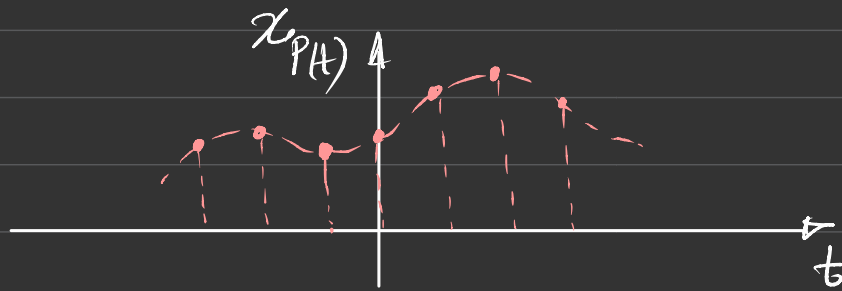


$$\omega_s = \frac{2\pi}{T}$$

$\omega_s > 2\omega_m$
 $\Rightarrow$ No overlapping



Ch. 7.2. RECONSTRUCTION FROM SAMPLES $x(nT)$

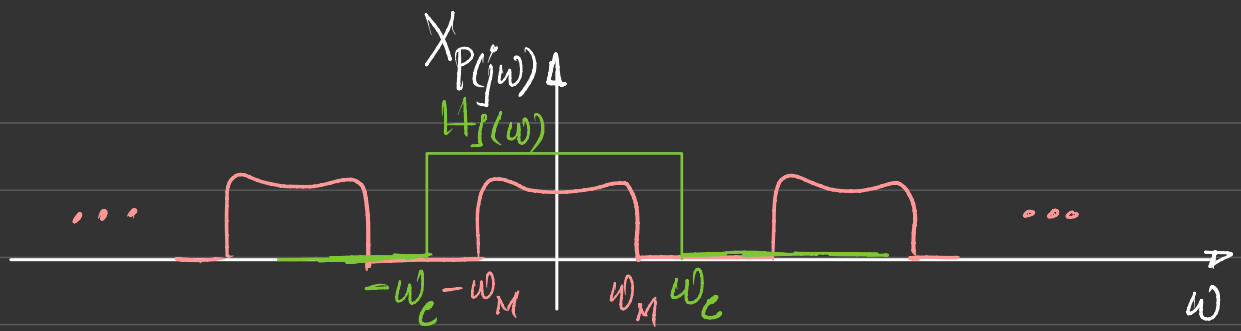


Interpolation

- Determination of "missing" part of a sig. b/w known parts.
- Fitting of a CT sig. to a set of sample values.

Ideal Interpolation with 'sinc function

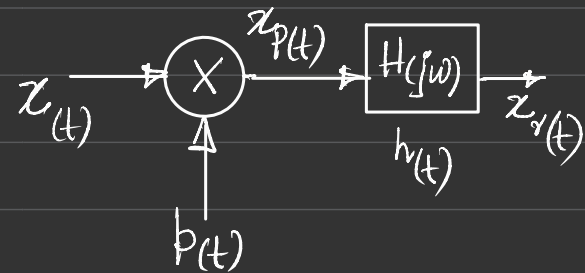
- For a band-limited sig. (limited by ω_m), if the sampling instants are sufficiently closed ($T \downarrow$, $f_s \uparrow = 1/T \downarrow$, $f_s \geq 2f_m$), then the sig. can be reconstructed exactly.
- That is, through the use of a low-pass filter, exact interpolation can be carried out b/w the sampling points.



$$H_I(j\omega) = \begin{cases} T & , \quad |\omega| < \omega_c \\ 0 & , \quad |\omega| > \omega_c \end{cases}$$

- To understand the reconstruction of $x(t)$ as a process of interpolation, let's look at the effect of LPF in time-domain.
- The o/p of the filter is:

$$X_r(j\omega) = X_P(j\omega) H_I(j\omega)$$



$$\begin{aligned} x(t) = x_r(t) &= x_P(t) * h_r(t) \\ &= \left(\sum_{n=-\infty}^{\infty} x(nT) \delta(t - nT) \right) * h_r(t) \\ &= \sum_{n=-\infty}^{\infty} x(nT) \underline{h_r(t - nT)} \quad - \textcircled{1} \end{aligned}$$

- Above eqn. describes how to fit a continuous curve b/w the sample points $x(nT)$ & consequently represents an interpolation formula.
- For ideal lowpass filter, represented by $H_I(j\omega)$, the impulse response, $h_r(t)$, is given as:

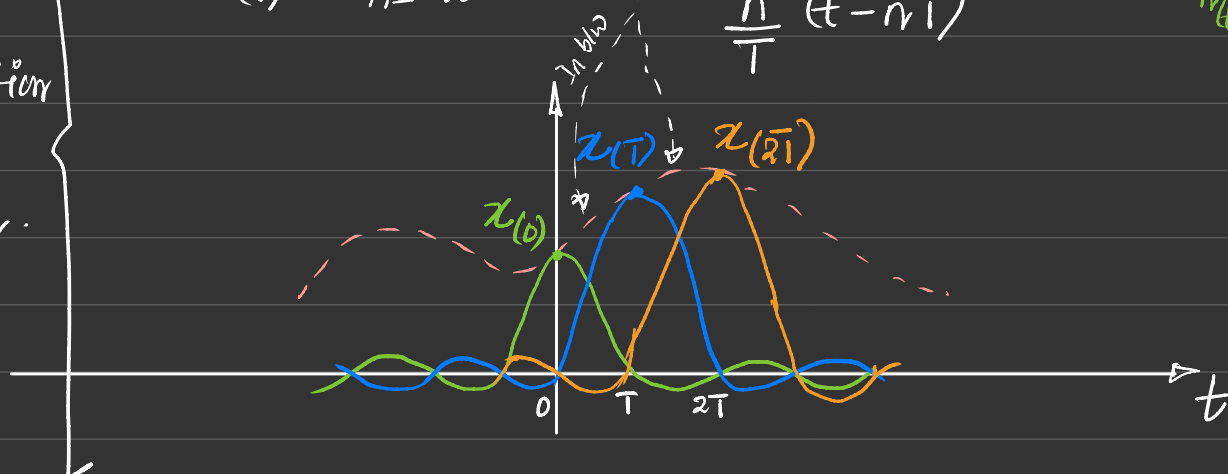
$$\begin{aligned} H_I(j\omega) &= \begin{cases} T & |\omega| < \omega_c \\ 0 & |\omega| > \omega_c \end{cases} \\ \updownarrow & \\ h_r(t) &= \frac{T \sin(\omega_c t)}{\pi t} \end{aligned}$$

- Usually, we use $\omega_c = \frac{\omega_s}{2} = \frac{2\pi/T}{2} = \frac{\pi}{T}$

- Accordingly,

$$x_r(t) = \sum_{n=-\infty}^{\infty} x(nT) \frac{\sin\left(\frac{\pi}{T}(t-nT)\right)}{\frac{\pi}{T}(t-nT)} \quad \text{--- (2)} \quad \text{Inserting } h(t) \text{ in eqn (1)}$$

Superposition
of sine
function.



- That's called "band-limited interpolation" [if $x(t)$ is band limited]

- In many cases, it is preferable to use a less accurate but simpler filter, or equivalently, a simpler interpolating function than the function in eqn (2).

- Interpolation with Zero-order Hold:

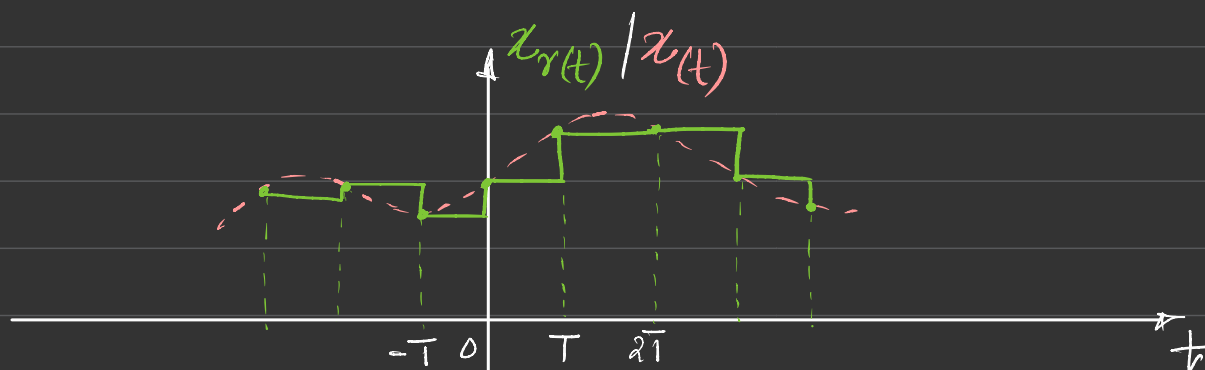
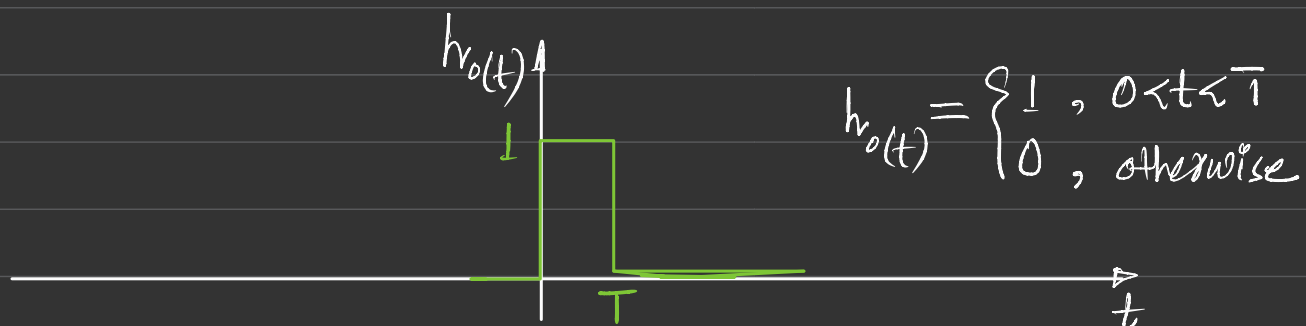


Fig. 1

$$x_r(t) = x(kT) \quad \text{for} \quad kT \leq t < (k+1)T$$

- "Zero-order hold" can be viewed as a form of interpolation b/w sample values in which the interpolating function $h(t)$ is the impulse response $h_0(t)$ given as:

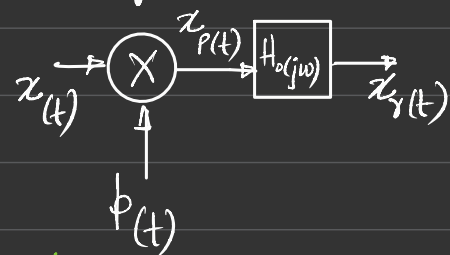


- $x_r(t)$ can mathematically be represented using $h_0(t)$ as:

$$x_r(t) = \sum_{n=-\infty}^{\infty} x(nT) h_0(t - nT)$$

$$= \sum_{n=-\infty}^{\infty} x(nT) (\delta(t - nT) * h_0(t))$$

$$= \underline{x_p(t)} * h_0(t)$$



- To view the spectra (for retrieval), take CTFT

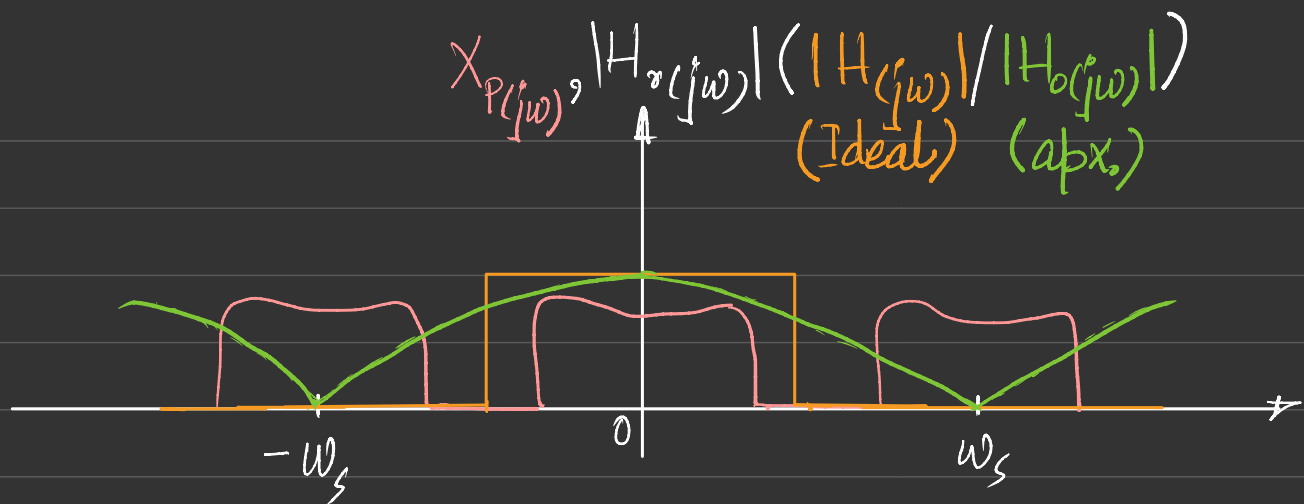
$$X_r(j\omega) = X_p(j\omega) H_0(j\omega)$$

where

$$H_0(j\omega) = e^{-j\omega T/2} \left(\frac{2 \sin(\omega T/2)}{\omega} \right)$$

Well known
shifted pulse
spectra.
(shifted by $T/2$)

$$|H_0(j\omega)| = \left| \frac{2 \sin(\omega T/2)}{\omega} \right|$$



- Comparing $X_r(j\omega)$ with $X(j\omega)$ (Spectrum of the original sig.)

with $H_r(j\omega)$ with $H_0(j\omega)$

- In case of $H_0(j\omega)$ (apx.), there is distortion in the main lobe.

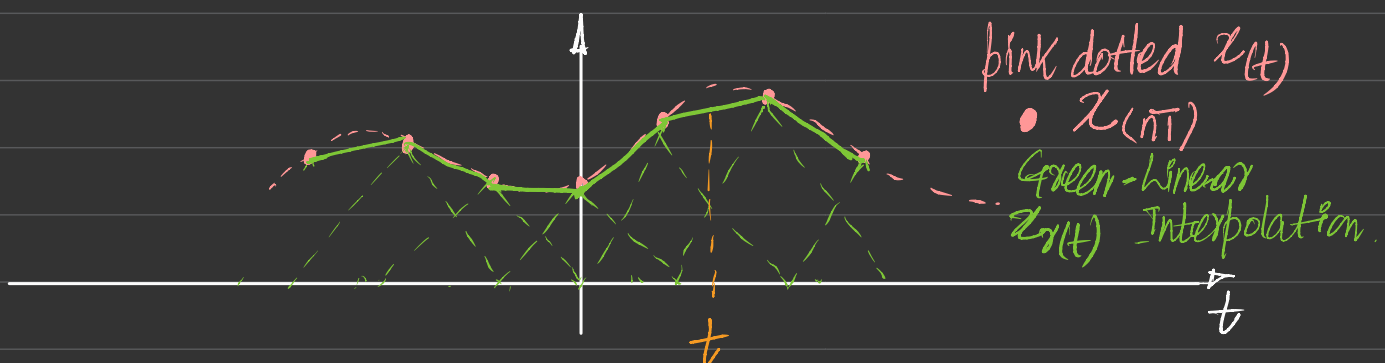
- Also, the side lobes pickup higher freq. noise.

- If the interpolation provided by the Zero-order hold is insufficient, we can use a variety of smoother interpolation strategies.

- In particular, the zero-order hold produces an output that is discontinuous (Fig. 1)

- In contrast, linear interpolation yields reconstructions that are continuous as described below.

- Linear Interpolation / Interpolation with First-order hold



- $x_r(t)$ is the weighted average of two samples.

$$x_r(t) = \frac{(k+1)T - t}{T} x_{(kT)} + \frac{(t - kT)}{T} x_{((k+1)T)}, \quad \forall kT \leq t < (k+1)T$$

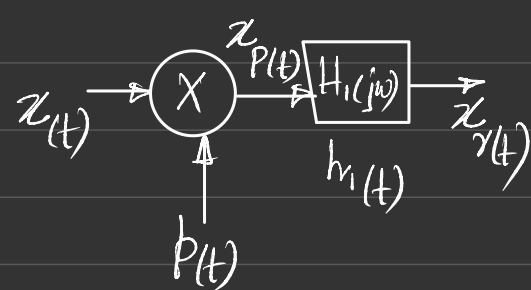
- The corresponding interpolation function is given as:



- $x_r(t)$ can be represented in terms

$$x_r(t) = \sum_{n=-\infty}^{\infty} x_{(nT)} h_1(t - nT)$$

$$= x_p(t) * h_1(t)$$

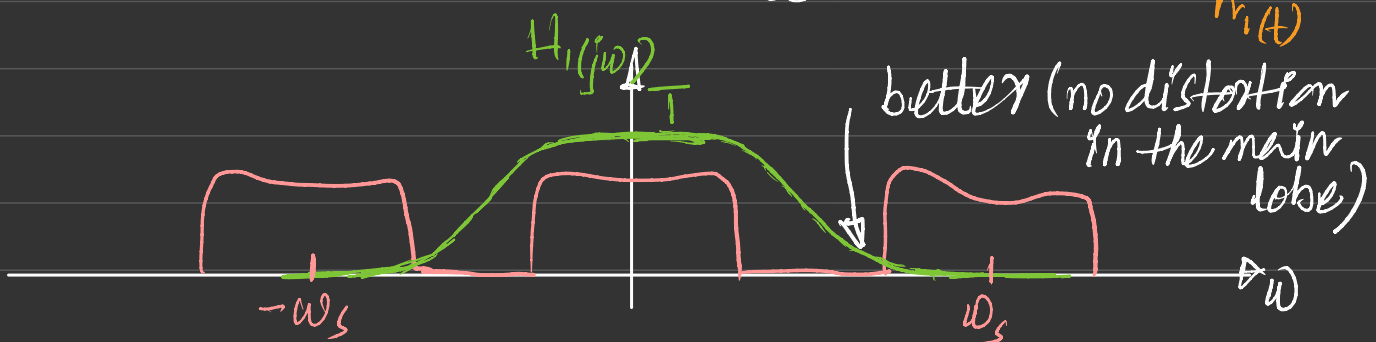


- For the spectra, take CTFT

$$X_r(j\omega) = X_p(j\omega) H_1(j\omega)$$

$$H_1(j\omega) = \frac{1}{T} \left(\frac{2 \sin(\omega T/2)}{\omega} \right)^2$$

✓ easy to prove for $h_1(t)$



- In an analogous fashion, we can define second & higher order filters to produce reconstructions with higher degree of smoothness.