

2 formulas u need to know

$$1) X_s(j\omega) = \frac{1}{T} \sum_{k=-\infty}^{\infty} X(j\omega - \frac{2\pi k}{T}) \quad FT \ x[n] \delta(n - NT)$$

$$2) X(e^{j\omega}) = X(j\frac{\omega}{T})$$

$$\omega = \Omega T$$

$$x_s(t) = \sum_n x[n] \delta(t - nT) \xrightarrow{FT} X_s(j\omega) = \sum_n x[n] e^{-jn\omega T} = X(e^{j\omega T})$$

$$\boxed{\begin{aligned} X(j\omega) &= X(e^{j\omega T}) \\ X_s(j\frac{\omega}{T}) &= X(e^{j\omega}) \end{aligned}}$$

$$\text{Sampling period } \Omega_s = \frac{2\pi}{T}$$

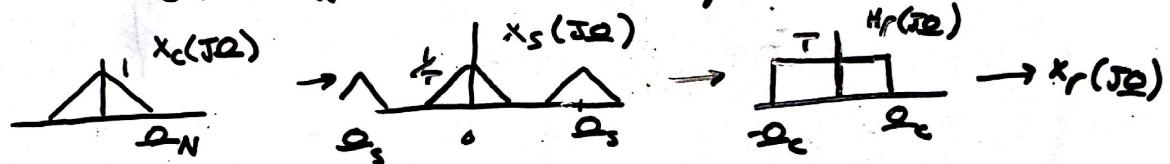
$$\begin{aligned} x[n] &\leftarrow x(e^{j\omega}) \rightarrow x[n] = x_c(nT) \\ x(t) &\leftarrow x(j\Omega) \quad x(e^{j\omega}) = x(e^{j\Omega T}) \\ x_s(t) &= x(nT) \leftarrow x(e^{j\Omega s t}) \quad \omega = \Omega T \end{aligned}$$

$$s(t) \leftarrow s(j\Omega) = \frac{2\pi}{T} \sum_{k=-\infty}^{\infty} \delta(\Omega - k\Omega_s)$$

$$x_s(t) \leftarrow x_s(j\Omega) = \frac{1}{T} \sum_{k=-\infty}^{\infty} x_c(j(\Omega - k\Omega_s))$$

$$x_c(j\Omega) = \begin{cases} 1 & \text{if } |\Omega| < \Omega_N \\ 0 & \text{else} \end{cases}$$

If  $\Omega_s \geq 2\Omega_N$  then no aliasing



If  $\Omega_N \leq \Omega_c \leq (\Omega_s - \Omega_N)$  then  $x_r(j\Omega) = x_c(j\Omega)$

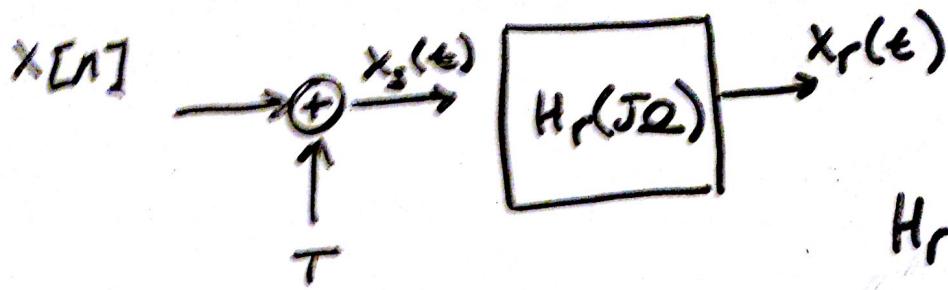
Nyquist-Shannon theorem:

$$x_s(j\Omega) = \sum_{n=-\infty}^{\infty} x_c(nT) e^{-jn\Omega T}$$

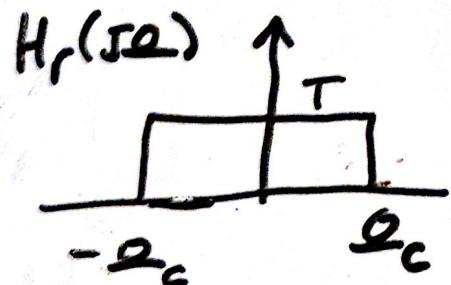
$$x_s(j\Omega) = x(e^{j\omega}) \Big|_{\omega=\Omega T} = x(e^{j\Omega s T})$$

$$x(e^{j\Omega s T}) = \frac{1}{T} \sum_{k=-\infty}^{\infty} x_c(j(\Omega - k\Omega_s))$$

### 4.3 reconstruction of bandlimited signal from its samples.



$$x_s(t) = \sum_{n=-\infty}^{\infty} x[n] \delta(t - nT)$$



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

$$X_r(j\Omega) = H_r(j\Omega) \sum_{n=-\infty}^{\infty} x[n] e^{-j\Omega nT}$$

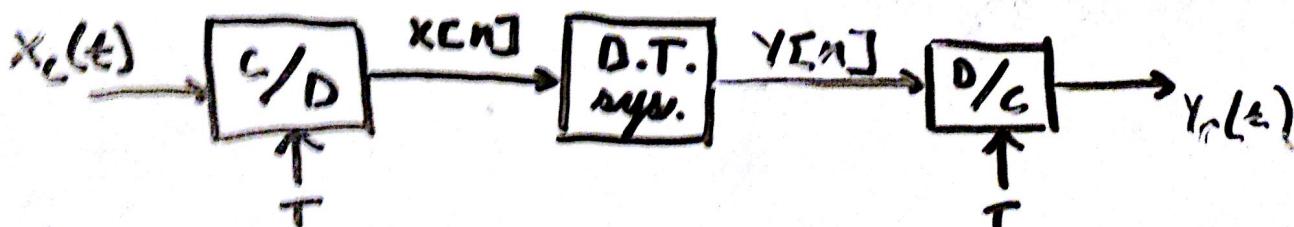
$$X_r(j\Omega) = H_r(j\Omega) X(e^{j\Omega T})$$

### 4.4 discrete time processing of cont. time signal

$$x[n] = x_c(nT)$$

$$X(e^{j\omega}) = \frac{1}{T} \sum_{n=-\infty}^{\infty} x[n] e^{-j\omega n} = \frac{1}{T} \sum_{k=-\infty}^{\infty} x_c(k - \Omega_s k)$$

$$Y_r(t) = \sum_{n=-\infty}^{\infty} Y[n] \frac{\sin(\pi(t - nT)/T)}{\pi(t - nT)/T}$$



$$4.4.1 \quad Y(e^{j\omega}) = H(e^{j\omega}) X(e^{j\omega})$$

$$Y_r(j\Omega) = H_r(j\Omega) H(e^{j\Omega T}) X(e^{j\omega})$$

$$Y_r(j\Omega) = H_r(j\Omega) H(e^{j\Omega T}) \frac{1}{T} \sum_{k=-\infty}^{\infty} x_c[j(\Omega - \frac{2\pi k}{T})]$$

$$Y_r(j\Omega) = \begin{cases} H(e^{j\Omega T}) X_c(j\Omega) & \text{if } |\Omega| < \frac{\pi}{T} \\ 0 & \text{else} \end{cases}$$

if  $X_c(j\Omega)$  is band limited and requires req:

$$\Omega_s \geq 2\Omega_N \iff \frac{2\pi}{T} \geq 2\Omega_N$$

$$Y_r(j\Omega) = H_{\text{eff}}(j\Omega) X_c(j\Omega) \text{ where } H_{\text{eff}}(j\Omega) = \begin{cases} H(e^{j\Omega T}) & \text{for } |\Omega| < \frac{\pi}{T} \\ 0 & \text{for } |\Omega| \geq \frac{\pi}{T} \end{cases}$$

4.6 Changing the sampling rate using discrete time processing

$$x[n] = x_c(nT)$$

reconstruction formula:  $x_r(t) = \sum_{n=-\infty}^{\infty} x[n] h_r(t-nT)$

$x_r[n] = x_c[nT]$  → change sampling rate

4.6.1

downsampling:



$$x_d[n] = x[nM]$$

if  $|\Omega| \geq \Omega_N$  then  $x_d[n]$  is

$$T_d = MT$$

$$\frac{2\pi}{T_d} = \frac{2\pi}{MT} \geq 2\Omega_N$$

$$X(e^{j\omega}) = \frac{1}{T} \sum_{k=-\infty}^{\infty} x_c[j(\frac{\omega}{T} - \frac{2\pi k}{T})]$$

similarly  $x_d[n] = x[nM] = x_c[n]$

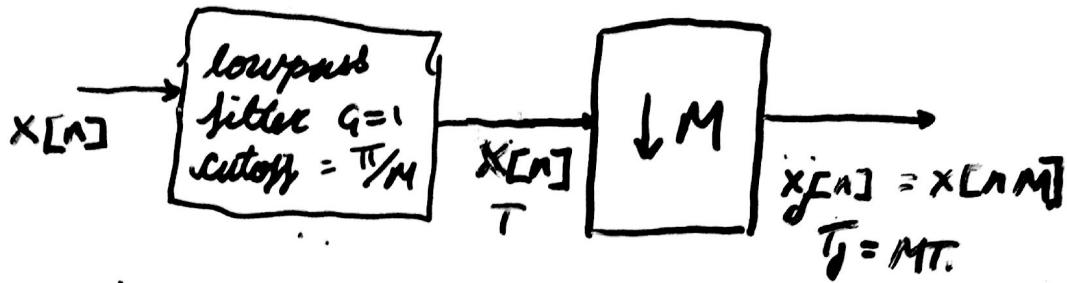
$$X_d(e^{j\omega}) = \frac{1}{M} \sum_{i=0}^{M-1} \left\{ \frac{1}{T} \sum_{k=-\infty}^{\infty} x_c[j(\frac{\omega}{MT} - \frac{2\pi k}{T} - \frac{2\pi i}{MT})] \right\}$$

\* we know this

$$\text{Eq(4.71)} \quad X(e^{j(\omega - 2\pi i)/M}) = \frac{1}{T} \sum_{k=-\infty}^{\infty} x_c[j(\frac{\omega - 2\pi i}{MT} - \frac{2\pi k}{T})]$$

$$\boxed{X_d(e^{j\omega}) = \frac{1}{M} \sum_{i=0}^{M-1} X(e^{j(\omega - 2\pi i/M)})}$$

general system for sampling rate reduction



#### 4.6.2 increasing sampling by integer

$$x_i[n] = x_c(nT_i)$$

upsample by factor  $L$

$$T_i = \frac{T}{L}$$

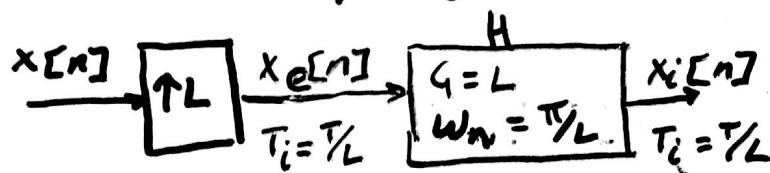
from sequence of samples

$$x[n] = x_c(nT)$$

$$x[n] = \begin{cases} x[n/L], & n = 0, \pm L, \pm 2L \dots \\ 0, & \text{else} \end{cases}$$

\* upsampling is same as interpolation

\* works if  $x[n]$  was obtained  $x[n] = x_c(nT)$  without aliasing



$$x_e[n] = \sum_{k=-\infty}^{\infty} x[k] \delta[n - kL]$$

$$x_e(e^{j\omega}) = \sum_{n=-\infty}^{\infty} \left( \sum_{k=-\infty}^{\infty} x[k] \delta[n - kL] \right) e^{-j\omega n} = \sum_{k=-\infty}^{\infty} x[k] e^{-j\omega L k} = x(e^{j\omega L})$$

- we have expanded the frequency scale
- $\omega = \omega L$  so  $\omega$  is now normalized by  $\omega = \Omega T_i$
- $\omega_n = \pi/L$
- $h_i[n] = \frac{\sin(\pi n/L)}{\pi n/L}$
- interpolation eq  $x_i[n] = x[k] \frac{\sin[\pi(n-kL)/L]}{\pi(n-kL)/L}$

# POST TEST 1

DSP lecture 9/23/15  
sampling theorem:

$x(t)$  = signal       $x_s(t)$  = sampled signal

$$x_s(t) = x(t) \sum_{n \in \mathbb{Z}} \delta(t - nt)$$

what happens to  $x_s(t)$  in Fourier?

can we recover  $x(t)$  from  $x_s(t)$

Q1:  $x_s(t)$  = Fourier transform       $x_s(t) = \int_{-\infty}^{\infty} X_s(\omega) e^{-j\omega t} d\omega$

$$X_s(j\Omega) = \frac{1}{2\pi} X(j\Omega) * S(j\Omega) \quad S - \text{for sampling}$$

where

$$S(t) = \sum_{n=-\infty}^{\infty} \delta(t - nT) \quad T - \text{sampling period}$$

find F.T.

$$S(j\Omega) = \sum_{n \in \mathbb{Z}} e^{-jn\Omega}$$

remember

$$\sum_{n=-\infty}^{\infty} e^{-jn\omega} = 2\pi \sum_{k=-\infty}^{\infty} \delta(\omega - 2\pi k)$$

if  $\omega = \Omega T$

$$\begin{aligned} \sum_{n=-\infty}^{\infty} e^{-jn\Omega T} &= 2\pi \sum_{k=-\infty}^{\infty} \delta(\Omega T - 2\pi k) \\ &= \frac{1}{T} \sum_{k=-\infty}^{\infty} \delta(\Omega - \frac{2\pi k}{T}) \end{aligned}$$

so

$$S(j\Omega) = \frac{2\pi}{T} \sum_{k=-\infty}^{\infty} \delta(\Omega - \frac{2\pi k}{T})$$

go back to  $X(j\Omega)$

\* when sampling,  
the F.T. must be  
periodic

$$X(j\Omega) = \frac{1}{2\pi} \sum_{k=-\infty}^{\infty} X(j\Omega - \frac{2\pi k}{T}) * \delta(\Omega - \frac{2\pi k}{T})$$

$$X(j\Omega) = \frac{1}{T} \sum_{k=-\infty}^{\infty} X(j\Omega - \frac{2\pi k}{T}) \leftarrow \frac{2\pi}{T} \text{ periodic}$$

What is the connection between

$$x_s(j\omega) = \frac{1}{T} \sum_k x(\tau(\omega - \frac{2\pi k}{T})) \rightarrow P = \frac{2\pi}{T}$$

and

$$X(e^{j\omega}) = \sum_n x[n] e^{-j\omega n} \rightarrow P = 2\pi$$

but

$$x_s(t) = \sum_{n=-\infty}^{\infty} \delta(t - nt) = \sum_{n=-\infty}^{\infty} x(nt) \delta(t - nt) \rightarrow x(t) @ nT$$

$$x_s(j\omega) = \sum_{n=-\infty}^{\infty} x(nT) e^{-jnT\omega} \rightarrow x[n] = x(nT)$$

$$x_s(j\omega) = \sum_{n=-\infty}^{\infty} x[n] e^{-jn\omega T} \rightarrow \begin{matrix} \omega = \omega T \\ \uparrow \quad \uparrow \\ C.T. \quad R_{\text{period}} \end{matrix}$$

Ex1:  $x(t) = \cos(4000\pi t)$  sample  $T = \frac{1}{6000} \text{ s}$   
what is  $x_s(j\omega)$ ?

F.T

$$\cos(4000\pi t) = \frac{e^{4000\pi t}}{2} + \frac{e^{-4000\pi t}}{2} \quad \omega = 4000\pi$$
$$\frac{2\pi}{T} = 12000\pi$$

$$X(j\omega) = \pi (\delta(\omega - 4000\pi) + \delta(\omega + 4000\pi))$$

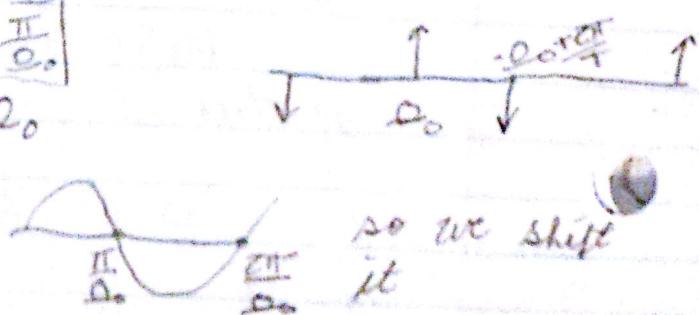
$$x_s(j\omega) = 6000\pi \sum_k \delta(\omega - 4000\pi - k\pi/12000) +$$
$$\delta(\omega + 4000\pi - k\pi/12000)$$

$$x(t) = \sin(\omega_0 t) \quad x(j\omega) = \frac{\pi}{\omega_0} (\delta(\omega - \omega_0) - \delta(\omega + \omega_0))$$

can we sample at exactly  $T = \frac{\pi}{\omega_0}$

$$-\omega_0 + \frac{2\pi}{T} \leq -\omega_0 + \frac{2\omega_0\pi}{\pi/\omega_0} = \omega_0$$

$$x_s(j\omega) \geq 0 \quad \text{in time,}$$



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$$\text{Ex 3: } x(t) = \frac{\sin(\omega_0 t)}{\omega_0 t}$$



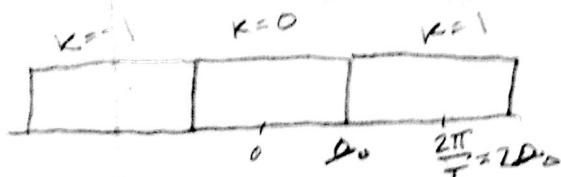
copy every  $\frac{2\pi}{T}$  so we want  $\frac{\pi}{T} = \omega_0$

$$\sin\left(\frac{\pi}{D_o} \cdot \omega_0\right) = \sin(\pi) = 0$$

@ n=0  $\frac{\sin(\frac{\pi}{D_o} \omega_0)}{\omega_0} = 1$ , the sampled function is  $\sin x$

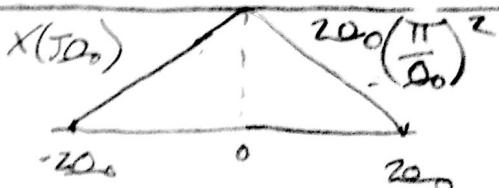


paradox: if we know  $x[n] = \delta[n]$  is a sampled at some period we do not expect that the original signal is a sinc



$$\frac{2\pi}{T} = 2\omega_0 \rightarrow \sum \delta[n] e^{jnw_n} = 1 \text{ for all } w$$

$$\text{Ex 4: } x(t) = \left[ \frac{\sin(\omega_0 t)}{\omega_0 t} \right]^2$$



if we sample every  $\frac{2\pi}{T}$ ,

$$T = \frac{\pi}{2\omega_0} \quad \text{and} \quad \frac{2\pi}{T} = 4\omega_0$$

What if we sample w/ larger period ( $\frac{\pi}{\omega_0}$ )  $\rightarrow$  smaller frequency  
→ aliasing.

$$x[n] = x\left(\frac{\pi}{D_o}\right) = \delta[n] + \delta[n-1]$$

reconstruction theorem (Nyquist):  
"Whistler 1915"

- signal must be bandlimited

$X(t)$  limited on  $[\omega_{\max}, \Omega_{\max}] \rightarrow X(j\Omega) = 0$

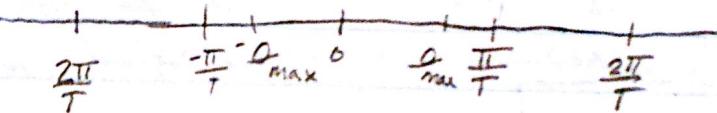
if  $|\Omega| > \Omega_{\max}$

$x(t)$  is uniquely characterized by the samples

@  $nT, n \in \mathbb{Z}$  where  $T \leq \frac{\pi}{\Omega_{\max}}$

sample fast enough

reconstruction:



we choose just the  $k=0$  term of the infinite sum.

filter:  $H_r(j\Omega) = \begin{cases} 1 & \text{if } |\Omega| \leq \frac{\pi}{T} \\ 0 & \text{else} \end{cases}$

$$x_s(j\Omega) = \frac{1}{T} \sum_{k=-\infty}^{\infty} X\left(\Omega - \frac{2\pi k}{T}\right)$$

$$x_s(j\Omega) H_r(j\Omega) = x(j\Omega) \rightarrow x_r(t) = h_r * x_s(t)$$

$$h_r(t) = \frac{\sin(\frac{\pi t}{T})}{\pi t}$$

$$x_s(t) = \sum x(nT) \delta(t - nT)$$

$$h_r * x_s = \sum_{n=-\infty}^{\infty} x(nT) \frac{\sin(\pi(t - nT)/T)}{\pi(t - nT)/T}$$

interpolation:

\* when you plug in a multiple of the period

you get the sample

$$= \sum x(nT) \frac{\sin(\pi(\frac{t}{T} - n))}{\pi(\frac{t}{T} - n)}$$

$$= \sum_{n=-\infty}^{\infty} x(nT) \frac{\sin(\pi(\frac{t}{T} - n))}{\pi(\frac{t}{T} - n)} \quad \text{if } t = mT$$

we get a  $\delta[m-n]$

observation

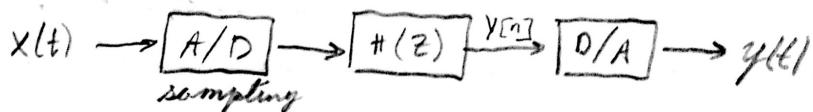
$$x(j\omega) H_r(j\omega) = x(j\omega)$$

↓

$$\text{DFT} \rightarrow x(e^{j\omega T}) H_r(j\omega) = x(j\omega)$$

$\stackrel{D2A}{\uparrow}$

discrete time implementation of analog filter



Question: what is the overall filter that maps  $x(t)$  to  $y(t)$   
 $y(t) = h_{\text{eff}} * x$

$$Y(e^{j\omega}) = H(e^{j\omega}) X(e^{j\omega})$$

$$X(e^{j\omega T}) = x_s(j\omega)$$

$$X(e^{j\omega T}) = \frac{1}{T} \sum_{k=-\infty}^{\infty} x(\omega - \frac{2\pi k}{T})$$

$$H_{\text{eff}}(\omega) = \begin{cases} H(e^{j\omega T}) & \text{if } |\omega| \leq \frac{\pi}{T} \\ 0 & \text{else} \end{cases}$$

$$\begin{aligned} Y(j\omega) &= H_r(j\omega) Y(e^{j\omega T}) \\ &= H_r(j\omega) H(e^{j\omega T}) X(e^{j\omega T}) \\ &= H_r(j\omega) H(e^{j\omega T}) \left( \sum_{k=-\infty}^{\infty} x(\omega - \frac{2\pi k}{T}) \right) \\ Y_1(j\omega) &= \begin{cases} H(e^{j\omega T}) \times (j\omega) & \text{if } |\omega| \leq \frac{\pi}{T} \\ 0 & \text{else} \end{cases} \end{aligned}$$

is

DSP lecture 9/28/15

HW PB 1  $x(t)$  bandlimitedsampling rate for recovery is  $T < \frac{1}{2}$ 

What if we sample  $\mathbb{Z}$  and  $\mathbb{Z} + a$  but  $T = 1$   
 $[0, 2\pi]$  and  $[-2\pi, 0]$  because you are breaking  
symmetry

$$x_s \text{ is } x_s(j\omega) = \sum x(\omega - 2\pi k)$$

$$x_a(j\omega) =$$

sample with a shifted train of impulses

$$s_a(t) = \sum_{n=-\infty}^{\infty} \delta(t - n - a) \quad \text{FT}\{x(t) s_a(t)\} = \frac{1}{2\pi} x(j\omega) * S_a(j\omega)$$

$$S_a(j\omega) = e^{-j\omega a} S(j\omega) \quad S(j\omega) = 2\pi \sum_k \delta(\omega - 2\pi k)$$

$$\text{F.T.}(x(t) s_a(t)) = x(j\omega) * e^{-j\omega a} \sum_k \delta(\omega - 2\pi k)$$

$\xrightarrow{* \sum_k e^{-j\omega a} \delta(\omega - 2\pi k)}$

~~property:~~

delta forces the function to be zero everywhere  
except at delta

$$\sum e^{-j\omega a 2\pi k} \delta(\omega - 2\pi k)$$

$$\text{F.T.}\{x(t) s_a(t)\} = \sum_{k=-\infty}^{\infty} e^{-jka 2\pi} x(\omega - 2\pi k)$$

$$[0, 2\pi] \quad x_s(j\omega) = x(j\omega) + x(j(\omega - 2\pi))$$

$$x_a(j\omega) = x(j\omega) + e^{-j2\pi a} x(j(\omega - 2\pi))$$

goal: solve for  $x(j\omega)$ 

$$\text{summary } x(j\omega) = A(j\omega) x_s(j\omega) + A(j\omega) x_a(j\omega)$$

$$A(j\omega) = \frac{1}{2\sin(\pi a)} \begin{cases} e^{j\pi(a-\frac{1}{2})} & \omega < 0 \\ e^{-j\pi(a-\frac{1}{2})} & \omega > 0 \end{cases}$$

$H(\Omega) = 1$  if  $|\Omega| < 2\pi$  claim for every  $\Omega \in [-\infty, \infty]$

$$x(\Omega) = H(\Omega) \{ A(\Omega) x_s(\Omega) + A^* x_a(\Omega) \}$$

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HW PB 2

$$\delta[n] \quad \frac{\sin \pi n}{\pi n}$$

prove  $\sum_{k=-\infty}^{\infty} F(\Omega - 2\pi k) = 1 \rightarrow$  same as  $f(e^{jw}) = 1$  then  $F[e^{j\omega}] = 1$  if  
F.T. of a sampled function

$$x_s(\Omega) = \frac{1}{T} \sum x(\Omega - \frac{2\pi k}{T})$$

$$T = 1$$

say  $f$  is the interpolation function

$$f[n] = \begin{cases} 1 & \text{if } n=0 \\ 0 & \text{else} \end{cases} \quad F(e^{jw}) = 1 \quad F[e^{j\Omega}] = f(j\Omega) = 1$$

~~W.W.~~

$$\begin{aligned} f(t) &\xrightarrow{\text{HARD}} f(t) s(t) = f_s(t) \rightarrow F_s(\Omega) = \frac{1}{T} \sum \\ &\xrightarrow{\text{D.T. easy}} \sum f[n] \delta(t - nT) \xrightarrow{\text{analog}} \sum f[n] e^{-j\Omega T n} \quad w = \Omega T \\ &\rightarrow F[e^{j\Omega}] \rightarrow F(e^{jw}) = \sum f[n] e^{-jwn} \end{aligned}$$

will be on final  $\rightarrow$  it cannot generate new frequencies

$$\text{LTI } \cos(\Omega_0 t) \rightarrow h(t) \in \mathbb{R} \rightarrow |H(\Omega)| \cos(\Omega_0 t + \angle H(\Omega))$$

$|H(j\omega_0)| \rightarrow$  system response at  $\omega_0$

$$\cos(\omega_0 t) \xrightarrow{H} |H(e^{j\omega_0})| \cos(\omega_0 t + \angle H(e^{j\omega_0}) + \angle)$$

## MULTIRATE SIGNAL PROCESSING

$$X(t) \rightarrow [A/D] \rightarrow X[n]$$

Here  $x[n] \rightarrow$  up/down sample

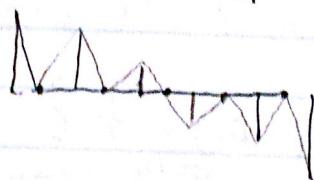
Change the sampling rate

→ understand in force in z-domains

Downsampling / decimation  $x \downarrow M$   $[n] = x[nM]$   $M$  integer  $\geq 1$   
 $\hookrightarrow$  linear but not time invariant

upSampling  $x \uparrow L[n] = \begin{cases} x[n/2] & \text{if } \frac{n}{2} \in \mathbb{Z} \\ 0 & \text{else} \end{cases}$

ex:  upsample by 2



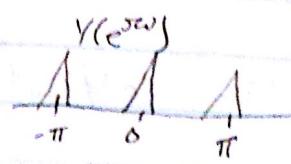
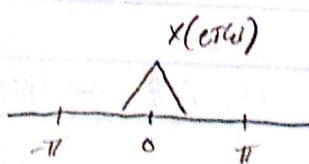
→ F.T. high frequencies

$$Y[n] = \begin{cases} x[n] & \text{if } e \\ 0 & \text{else} \end{cases}$$

$$V(z) = \sum_{m=-\infty}^{\infty} x[m] z^{-m}$$

$$= \sum_{n=-\infty}^{\infty} X[n] (z^2)^{-n} = X(z^2)$$

$$\text{let } L = 2 \quad z = e^{j\omega} \\ Y(e^{j\omega}) = X(e^{j2\omega})$$



\* never pass " upsample w/o low

↳ interpolate < moving avg >

## Z Transform of downsampled signal

$$M=2 \quad Y(z) = \sum_{n=-\infty}^{\infty} y[n] z^{-n} = \sum_{n=-\infty}^{\infty} x[2n] z^{-n}$$

$$Y(z^2) = \sum_{n=-\infty}^{\infty} x[2n] z^{-2n} \rightarrow x(z) = \sum x[2n] z^{-2n} + x[2n+1] z^{-2n-1}$$

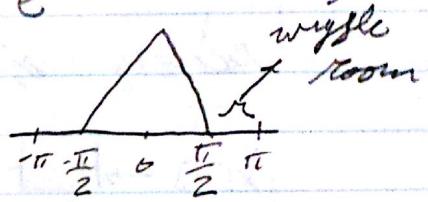
want to cancel odd

$$Y(z^2) = \frac{1}{2}(x(z) + x(-z)) \rightarrow x(z) + x(-z) = 2 \sum_{n=-\infty}^{\infty} x[2n] z^{-2n}$$

$$Y(z) = \frac{1}{2}[x(z^2) + x(-z^2)] \rightarrow \text{for } z = e^{jw}$$

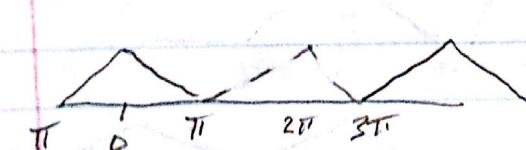
$$Y(e^{jw}) = \frac{1}{2}[x(e^{jw/2}) + x(e^{j(w-2\pi)/2})]$$

↑  $\frac{w}{2\pi}$  periodic



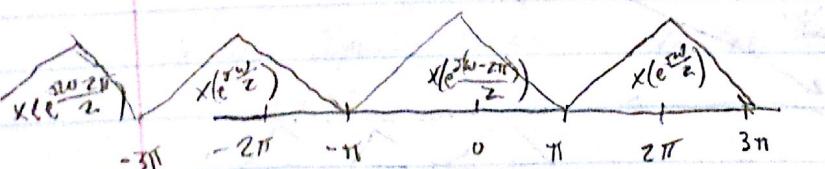
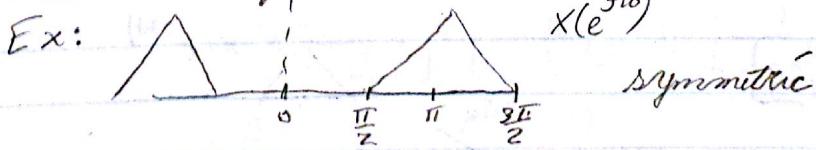
$$Y(t) = \sum_n x(t-nT) \rightarrow \text{periodic on } T$$

delay and copy



→ if there was no wiggle room we would get aliasing

→ the key is bandwidth → if  $BW < \pi$  you can downsample



$$M=3; \quad Y(z) = \frac{1}{m} \sum_{k=0}^{M-1} X\left(e^{-\frac{j2\pi k}{m}} z^{\frac{1}{m}}\right)$$

Note  $e^{j\frac{2\pi k}{m}} = 1$

$$Y(e^{jw}) = \frac{1}{m} \sum_{k=0}^{M-1} X\left(e^{j\frac{(w-2\pi k)}{m}}\right)$$

$$EX: M=3 \quad Y(e^{j\omega}) = \frac{1}{3} \sum_{k=0}^{3-1} X(e^{\frac{j(\omega - 2\pi k)}{3}})$$

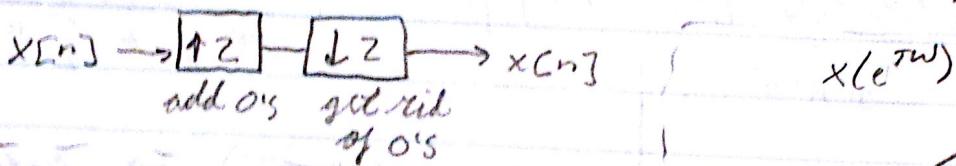
No aliasing if  $BW < \frac{\pi}{3}$

$$Y(e^{j\omega}) = \frac{1}{3} \left[ X(e^{j\omega/3}) + X(e^{\frac{j(\omega - 2\pi)}{3}}) + X(e^{\frac{j(\omega - 4\pi)}{3}}) \right]$$

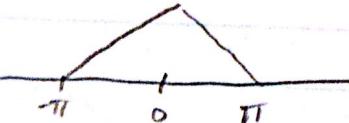
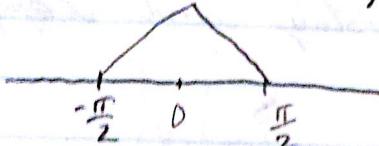
$\frac{T}{2} = 6\pi$   
 $BW = 2\pi$

You need 2 delays to make  $y(e^{j\omega})$  periodic on  $2\pi$   
 shift on factors of  $2\pi$  to stay periodic

Recovery of a signal after upsampling



$$Y(e^{j\omega}) = X(e^{j\omega/2})$$



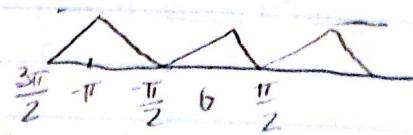
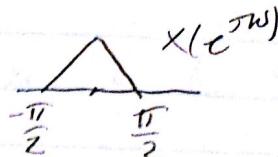
$$x(e^{j\omega/2})$$

Shift and add to recover original amplitude

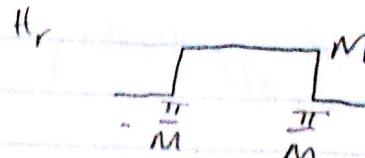
Recover after downsample

$$x[n] \rightarrow [ \downarrow 2 ] \rightarrow [ \uparrow 2 ] \rightarrow$$

$$Y(e^{j\omega}) = X(e^{j\omega/2})$$

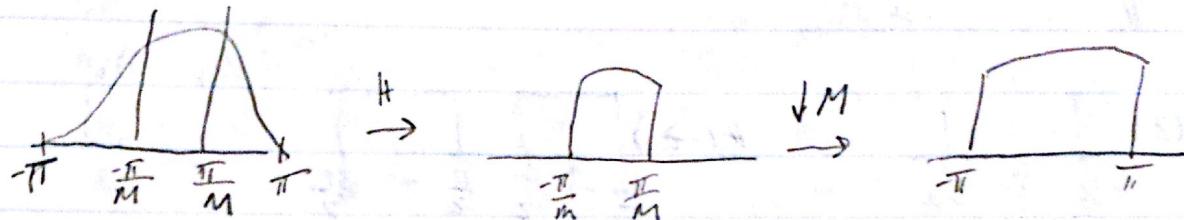
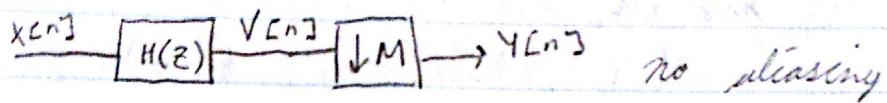


need Low Pass

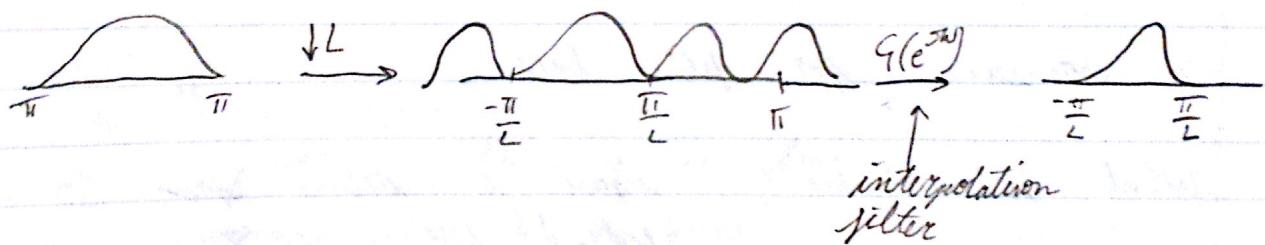
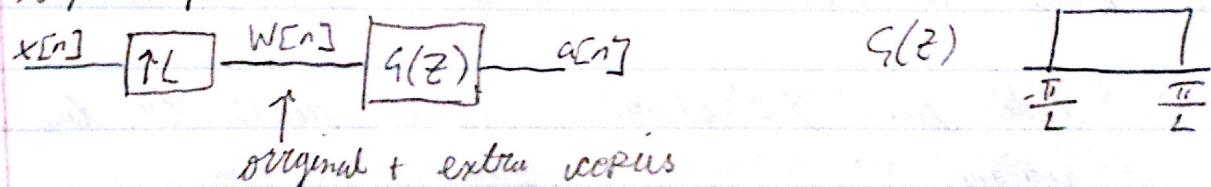


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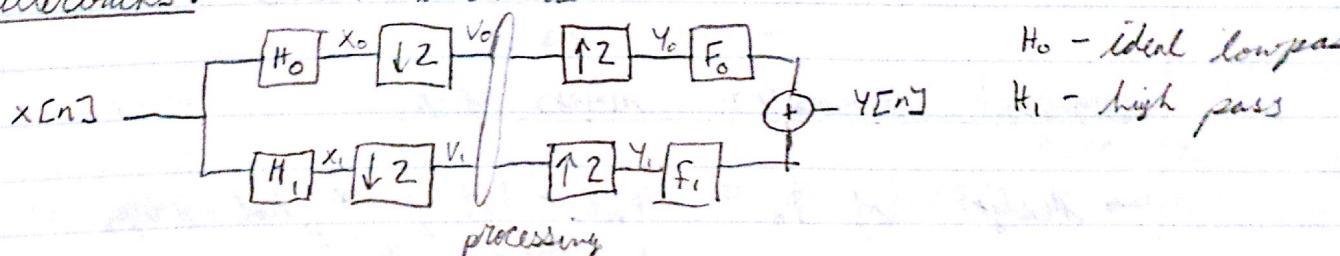
downsample and upsampling filters



upsample



filterbacks: max decimated



finite impulse response filter  $\rightarrow$  work in  $z$  domain  
 $i \in 0, 1$

$$x_i(z) = H_i(z) X(z)$$

$$v_i(z) = \frac{1}{2} \{ x_i(z^{1/2}) + x_i(-z^{1/2}) \}$$

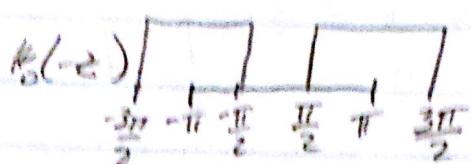
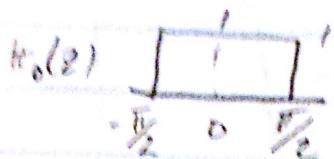
$$y_i(z) = V_i(z^2) = \frac{1}{2} \{ x_i(z) + x_i(-z) \} = \frac{1}{2} \{ H_i(z) X(z) + H_i(z) X(-z) \}$$

$$Y(z) = f_0(z) Y_0(z) + f_1(z) Y_1(z)$$

$$Y(z) = \frac{1}{2}x(z)\{f_0(z)H_0(z) + f_1(z)H_1(z)\} + \frac{1}{2}x(-z)\{f_0(-z)H_0(-z) + f_1(-z)H_1(-z)\}$$

if  $x(z) = Y(z)$  we need  $\{f_0(z)H_0(z) + f_1(z)H_1(z)\} = 0 \forall z$   
 $\{f_0(-z)H_0(-z) + f_1(-z)H_1(-z)\} = 0 \forall z$

1.  $H_0 = f_0, H_1 = f_1$



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HW code can be shared. Report needs to be unique

2 extensions for filter back

What is extension? signal is defined from  $[0, \pi]$   
 extend from  $0 \rightarrow -\pi$  as a  
 mirror symmetric

conjugate quadrature mirror filter

→ Design of  $H_0$   $H_1(z) = H_0(-z)$  is not helpful

$$H_1(z) = -z^{2K^H} H_0(-z^2)$$

①  $H_1(z) = -z^N H_0(-z^2)$   $N$ : odd  $N$ : is the size of the filter

② in addition  $H_0(z)H_0(z^2) + H_0(-z)H_0(-z^2) = 2$

③ Aliasing cancellation  $f_0(z) = H_0(z)$   $f_1(z) = -H_0(z^2)$

if  $H_0, f_0$  and  $f_1$  are constructed using ①, ②, ③  
 then  $y(z) = z^N x(z)$   $y[n] = x[n-N]$   
 $\uparrow$   
 output of filter

interpretation of ②:

$$H(z) H_o(z^{-1}) + H_o(-z) H_o(-z^{-1}) = 2$$

$$H_o(e^{jw}) H_o(e^{-jw}) + H_o(e^{j(w-\pi)}) H_o(e^{-j(w-\pi)}) = 2$$

$$H_o(e^{jw}) = \sum_{n=0}^{\infty} h_o[n] e^{jn\omega} = [H_o(e^{jw})]^*$$

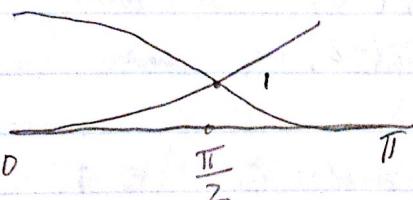
Note  $H_o$  is real

so

$$|H_o(e^{jw})|^2 + |H_o(e^{j(w-\pi)})|^2 = 2$$

$$\left| H_o(e^{j(\frac{\pi}{2}-\lambda)}) \right|^2 + \left| H_o(e^{j(\frac{\pi}{2}+\lambda)}) \right|^2 = 2 \quad \begin{matrix} [\lambda, \pi] \\ \downarrow \\ w = \frac{\pi}{2} + \lambda \end{matrix}$$

$|H_o(e^{jw})|$  is even because  $h_o[n]$  is real  
at  $w = \frac{\pi}{2}$  the  $\Re[H_o(e^{j\frac{\pi}{2}})] = 2$



$$|H_o(e^{jw})|^2 = \cos^2(w/2)$$

$$\cos^2(w - \pi/2) = \sin^2(w/2)$$

solution

$$H_o(e^{jw}) = \sqrt{2} \cos(wt)$$

Euler  $e^{-jw/2} + e^{jw/2} \rightarrow$  in time  
time domain, ① ② ③ yield:

$$h_o[n] = (-1)^n h_o(2k-1-n)$$

↑ time reversal

$$H_o(z) = -z^{-2k+1} H_o(-z^{-1})$$

$$= -z^{-2k+1} \{ h_o[n] (-1)^n \}$$

$$= \sum_n h_o[n] (-1)^{n-1} z^{-(2k-1-n)}$$

$$F_o[n] = h_o[2k-1-n]$$

$$F_o[n] = h_o[2k-1-n]$$

→ problem: construction prevents the filter to have linear phase.

→ they cannot be symmetric in time

→ delays based on frequency

→ solution: use spline filter

filter back using spline filter:  $H_0$  and  $F_0$  are needed

→ generalize the previous construction to allow linear phase

→ more degrees of freedom: carry 2 filters  $H_0, F_0$  analysis  $\nwarrow$  reconstruction

replace eq (1)(2)(3) by new equations

$$H_0(z) H_0(z^{-1}) + H_0(-z) H_0(-z^{-1}) = z \Rightarrow H_0(z) f_0(z) + H_0(-z) f_0(-z) = z$$

$$H_0(z) = H_0(-z) \Rightarrow H_0(z) = z^N F_0(-z) \quad N: \text{odd}$$

$$F_0(z) = -H_0(-z) \Rightarrow F_0(z) = z^N H_0(-z)$$

if we can design  $H_0$  and  $F_0$  S.T. (1), (2), (3)  
hold then output of filter back is a pure delay  
 $y(z) = z^N x(z)$  ← pure delay

plug and look for powers of  
linear phase for all symmetric signals

DSP lecture 10/12/15

## z-transform analysis of digital filters

identify decimate properties of a filter

frequency response:

$$\xrightarrow{H(z)} Y(z) \quad z\text{-domain} \quad Y(z) = H(z)X(z)$$

$$\text{on unit circle } Y(e^{j\omega}) = H(e^{j\omega})X(e^{j\omega})$$

$$\text{magnitude } |H(e^{j\omega})|, \angle H(e^{j\omega})$$

effect of phase on  $y[n]$ ?

→ consider "narrow" band signal  $x[n] = A \cos(\omega_0 n + \phi)$

$$y[n] = A |H(e^{j\omega})| \cos(\omega_0 n + \phi + \angle H(e^{j\omega}))$$

→ if the phase is a linear function of the frequency,  
then there is  $\tau$  s.t.  $\angle H(e^{j\omega}) = -\omega\tau + \gamma_0$

$$\begin{aligned} \rightarrow y[n] &= |H(e^{j\omega})| A \cos(\omega_0 n + \phi - \omega_0 \tau + \gamma_0) \\ &= |H(e^{j\omega})| A \cos((\omega_0(n-\tau) + \phi + \gamma_0)) \end{aligned}$$

↑ time shift/delay

if  $\gamma_0 = 0$   
not a constraint

linear phase: time shift/delay that does not depend  
on  $|\omega|$   
super important in audio processing

ex: single zero/ single pole

$$H(z) = 1 - \alpha_0 z^{-1} = \frac{z}{z} (1 - \alpha_0 z^{-1}) = \frac{1}{z} (z - \alpha_0) \rightarrow \text{zero at } \alpha_0$$

$$\alpha_0 = r e^{j\theta} \quad H(e^{j\omega}) = 1 - (r e^{j\theta} e^{j\omega}) = 1 - r e^{j(\omega+\theta)}$$

$$|H(e^{j\omega})|^2 = 1 + r^2 - 2r \left( e^{j(\omega+\theta)} + e^{-j(\omega+\theta)} \right) = 1 + r^2 - 2r \cos(\omega+\theta)$$

H.P.

O.O

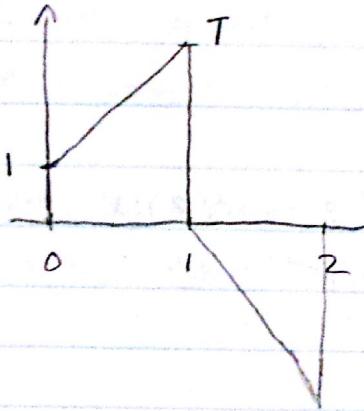
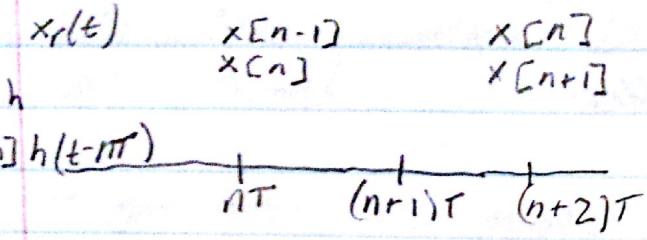
P.I

O.I

C.I

2

DSP lecture 10/15/15 lecture before test 2



$$\text{on } (nT, (n+1)T) \quad 1: x[n] + \frac{1}{T}(x[n](t-nT)) \\ = x[n] \left\{ 1 + \frac{1}{T}(t - nT) \right\} \\ = x[n] h[t - nT]$$

on  $(nT, (n+2)T)$ :

the part of  $x_r(t)$  that contains  $x[n] - x[n] \frac{1}{T}(t - (n+1)T)$

Q: where did the minus sign come from

$$x_r(t) = x[n] + \frac{x[n] - x[n]}{T} (t - (n+1)T)$$

\* you want to estimate slope so you borrow the previous term

$$H(j\omega) = \int_{-\infty}^{\infty} h(t) e^{-j\omega t} dt = \int_0^1 (1 + \frac{t}{T}) e^{-j\omega t} dt + \int_1^2$$

→ integration by parts

$$\int_0^1 t e^{-j\omega t} dt = \frac{t e^{-j\omega t}}{-j\omega} \Big|_0^1 + \frac{1}{-j\omega} \int_0^1 e^{-j\omega t} dt$$

$$H(j\omega) = T(1 + j\omega T) \left[ \frac{\sin(\omega T/2)}{\omega T/2} \right]^2$$

\* not the sinc because filter is not perfect

complete angle

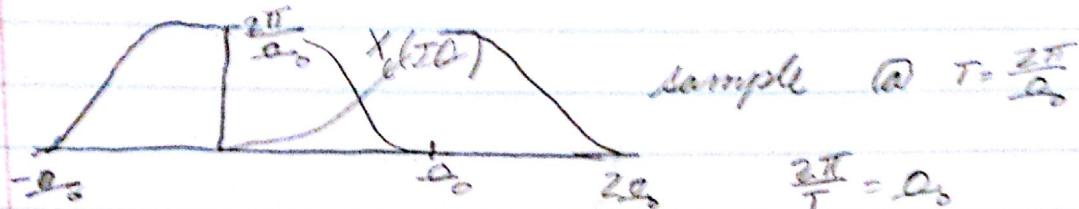
$$\therefore H(j\omega) = T(1 + j\omega T) = \text{rectm}(\omega T)$$

look up zero order hold

problem 4

$$x_c(t) = ?$$

$$x(j\omega) = \frac{\pi}{\omega_0} (1 + \cos(\frac{\pi}{\omega_0} \omega)) \quad \text{if } |\omega| \leq \omega_0.$$



$$x_c(t) = f(t - nT) \rightarrow \text{what is the constant? is a delta}$$

$$1 + \cos(2\theta) \rightarrow \text{what is } \cos(2\theta) = \cos^2(\theta) - \sin^2(\theta) \\ = 2\cos^2(\theta) - 1$$

$$\frac{1 + \cos(2\theta)}{2} = \cos^2(\theta) \text{ so } 1 + \cos(2\theta) \rightarrow 2\cos^2(\theta)$$

$$x(j\omega) = \frac{\pi}{\omega_0} 2\cos^2\left(\frac{\pi\omega}{2\omega_0}\right)$$

$$x_c(j\omega_0) = \frac{\pi}{\omega_0} \left( 2\cos^2\left(\frac{\pi(\omega - \omega_0)}{2\omega_0}\right) \right) = \cos^2\left(\frac{\pi}{2\omega_0} - \frac{\pi}{2}\right) = \sin^2\left(\frac{\pi}{2\omega_0}\right)$$

problem 5

$$|x(j\omega)| = 0 \quad \text{if } |\omega| \geq \omega_m$$

$$y(t) \leq x(t) \sum_{n=1}^{\infty} \underbrace{\lambda^n \sin(n\omega_0 t)}_{a(t)}$$

→ take x and move away from zero

$$y(j\omega) = \frac{1}{\pi} x \neq A(j\omega)$$

→ if \omega\_0 is large there is no cross talk between copies

$$A(j\omega) = \varepsilon \lambda^n \pi (\delta(\omega - n\omega_0) - \delta(\omega + n\omega_0)) \rightarrow \text{A.M.}$$

