

Last Class:

* Fast Fourier Transform (FFT)

- * direct DFT - $O(N^2)$ multiplications & additions
- * radix-2 FFT - $O(N \log_2 N)$ mult. & additions, $N = 2^m$
 - efficient implementation based on computing shorter DFTs recursively & using properties of phase factors W_N .
- * we saw decimation-in-time approach.

Today's class:

Some remarks on FFT:

- Inverse DFT has almost identical efficient implementation in $O(N \log_2 N)$.
- verify by looking at the IDFT formula.
- We have radix-4 algorithm as well (& other variations)
- In general when N is composite we can use divide & conquer approach i.e. Cooley-Tukey algorithm (& other variations)
- Decimation-in-time requires "bit-reversed" input ordering (of time)
- We also have decimation-in-frequency approach (radix-2)

$$X[k] = \sum_{n=0}^{N-1} x[n] W_N^{kn} \quad \text{say } N = 2^m.$$

$$k \text{ - even : } X[2r] = \sum_{n=0}^{N-1} x[n] W_N^{2rn}, \quad k = 2r$$

$$k \text{ - odd : } X[2r+1] = \sum_{n=0}^{N-1} x[n] W_N^{(2r+1)n}, \quad k = 2r$$

$$X[2r] = \sum_{n=0}^{N-1} x[n] W_{N/2}^{rn} = \sum_{n=0}^{N/2-1} x[n] W_{N/2}^{rn} + \sum_{n=0}^{N/2-1} x[n + \frac{N}{2}] W_{N/2}^{rn}$$

$$\therefore \quad \sum_{n=0}^{N/2-1} (x[n] + x[n + \frac{N}{2}]) W_{N/2}^{rn}, \quad r = 0, 1, \dots, \frac{N}{2}-1$$

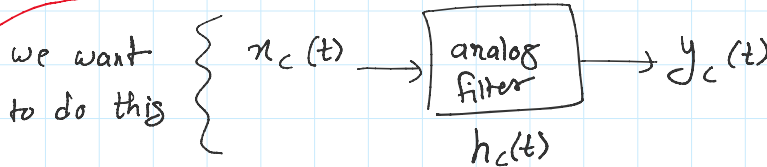
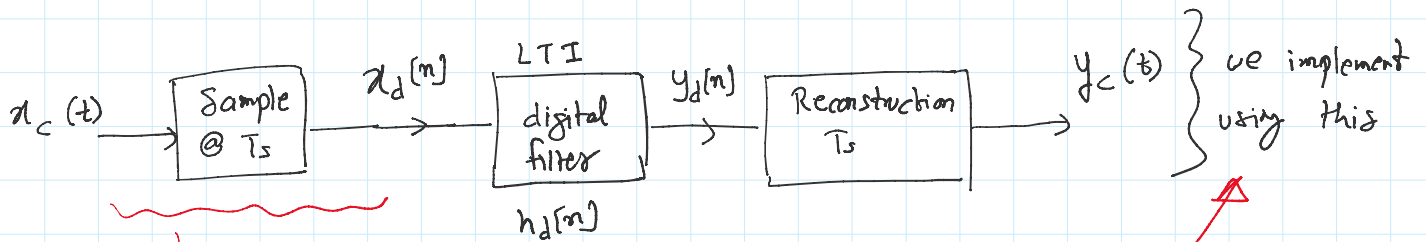
$$\text{also, } W_{N/2}^{r(n+N/2)} = W_{N/2}^{rn}$$

$$X[2r] = \sum_{n=0}^{N/2-1} (x[n] + x[n + \frac{N}{2}]) W_N^{rn} \quad r = 0, 1, \dots, \frac{N}{2}-1$$

$\frac{N}{2}$ -point DFT of $y[n] = x[n] + x[n + \frac{N}{2}]$, $n = 0, 1, \dots, \frac{N}{2}-1$.

(f) performing convolution using FFT. - can be done efficiently in $O(N \log_2 N)$ whereas direct convolution sum would be $O(N^2)$.

★ Processing analog signals using digital filters. ★

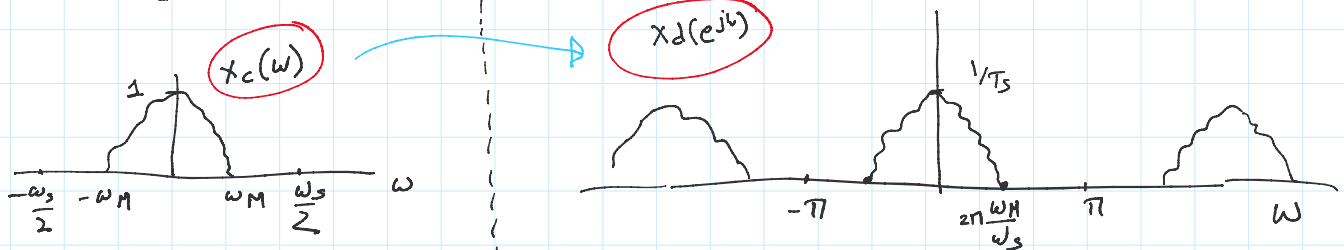


$$\omega_s = \frac{2\pi}{T_s}$$

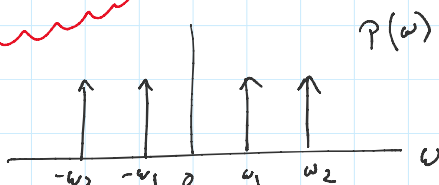
$$x_c(t) \xleftrightarrow{\text{FT}} X_c(\omega)$$

$$x_d[n] \xleftrightarrow{\text{DTFT}} X_d(e^{j\omega}) = \frac{1}{T_s} \sum_{k=-\infty}^{\infty} X_c((\omega - k\omega_s)T_s) \quad \text{★}$$

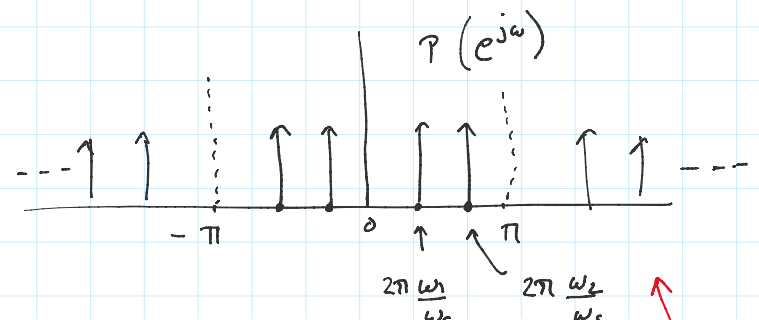
$x_c(nT_s)$



LAB: 8.1



$$\cos(\omega_1 t) + \cos(\omega_2 t)$$



$$\cos(\omega_1 t) + \cos(\omega_2 t)$$

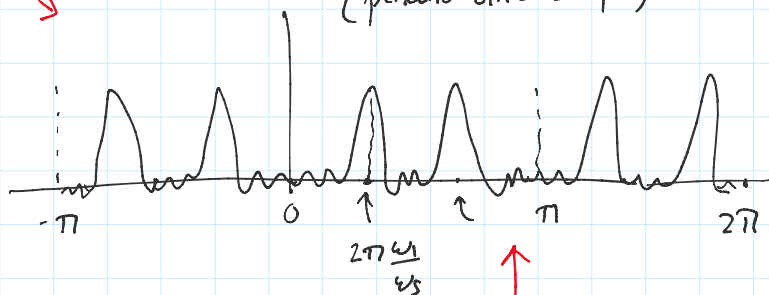
$$p[n] = \cos(\omega_1 n T_s) + \cos(\omega_2 n T_s)$$

$$= \cos\left(2\pi \frac{\omega_1}{\omega_s} n\right) + \cos\left(2\pi \frac{\omega_2}{\omega_s} n\right)$$

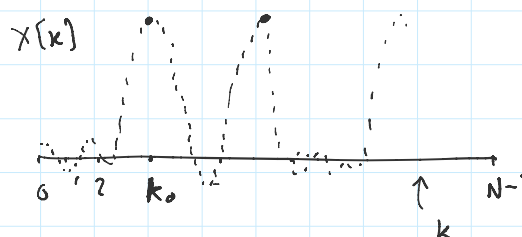
$$x[n] = p[n] \cdot w[n]$$

$$X(e^{j\omega}) = \frac{1}{2\pi} [P(e^{j\omega}) * W(e^{j\omega})]$$

for rectangular
(periodic sinc shape)



DFT: $X[k]$ i.e.
samples of $X(e^{j\omega})$



k_0 corresponds to freq. $2\pi \frac{k_0}{N}$

1st peak at k_0

$$\Rightarrow 2\pi \frac{\hat{\omega}_1}{\omega_s} = 2\pi \frac{k_0}{N}$$

$$\Rightarrow \hat{\omega}_1 = \frac{k_0 \omega_s}{N} \quad \text{estimate}$$

Tone $\omega_1 = \frac{\omega_s}{8}$

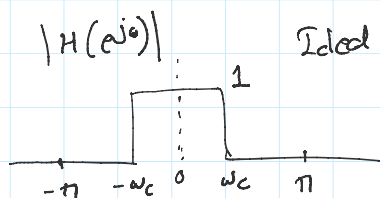
* Digital Filter Design *

Ideal filters:

$$x[n] \rightarrow \boxed{h[n]} \rightarrow y[n] = x[n] * h[n]$$

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

filter response



Ideal LPF $\longleftrightarrow h[n]$

$$h[n] = \frac{\sin(\omega_c n)}{\pi n}$$

not practical because

- infinite length
- not causal.

★ We focus on digital filters of the form:

$$\frac{y(z)}{x(z)} = H(z) = \frac{b_0 + b_1 z^{-1} + \dots + b_M z^{-M}}{1 + a_1 z^{-1} + \dots + a_N z^{-N}} = \frac{N(z)}{D(z)}$$

difference equation relation between $x[n]$ & $y[n]$.

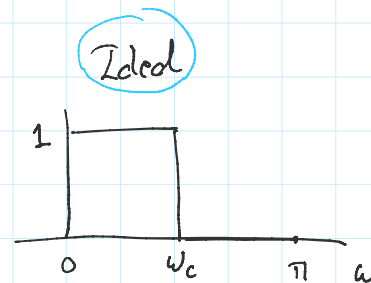
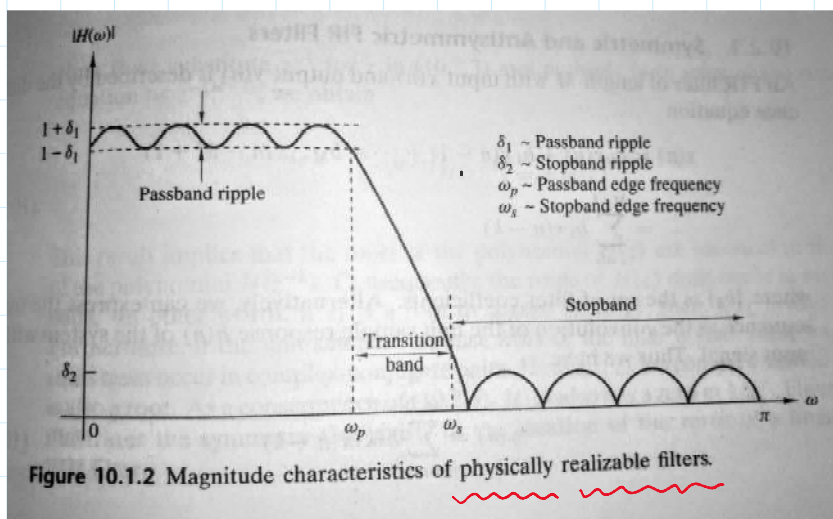
$$y[n] = \sum_{k=0}^M b_k x[n-k] - \sum_{l=1}^N a_l y[n-l] \quad \left. \vphantom{\sum_{k=0}^M b_k x[n-k]} \right\} \text{ can be implemented with finite computation.}$$

Cases (a) $a_i = 0 \quad \forall i = 1, 2, \dots, N.$

\Rightarrow Finite Impulse Response (FIR) filters

(b) otherwise: Infinite impulse response (IIR) filters

★ Filter design: given $H_D(e^{j\omega})$ i.e. desired filter find $h[n]$ so that its DTFT is close to $H_D(e^{j\omega})$.



In practice.

← from proakis