

UNIVERSITY OF ILLINOIS AT URBANA-CHAMPAIGN  
Department of Electrical and Computer Engineering

ECE 310 DIGITAL SIGNAL PROCESSING  
Spring 1999

**EXAM 3**

Wednesday, April 21, 1999

- This is a **CLOSED BOOK** exam, but three  $8\frac{1}{2}'' \times 11''$  sheet of notes (both sides) are allowed (no xeroxed sheets).
- **NO CALCULATORS** are allowed.
- There are 4 problems on the exam, with point values indicated, and not in order of difficulty.
- You must **SHOW YOUR WORK** to get full credit.
- Please **BE NEAT**—we can't grade what we can't decipher.
- Don't forget to **PUT YOUR NAME ON ALL SHEETS**.
- Only this booklet is handed in. **NOTHING ELSE WILL BE CONSIDERED.**

Problem	Score
1	
2	
3	
4	
Total	

Name: \_\_\_\_\_

Circle One:    LEVINSON    SINGER

**Problem 1 (10 points)**

Consider the system in the figure below. Let  $x_a(t)$  be an analog signal bandlimited to  $\frac{\pi}{T}$ , i.e.,

$$X_a(\Omega) = 0, \quad |\Omega| > \frac{\pi}{T}$$

This signal is sampled at a rate of  $1/T$  samples per second, such that  $x[n] = x_a(nT)$ . The signal  $x[n]$  is processed by the discrete-time system with impulse response  $h[n]$ , such that

$$y[n] = \frac{1}{2}y[n-1] + x[n].$$

Next, the signal  $y[n]$  is passed through a zero-order hold (ZOH) device, such that

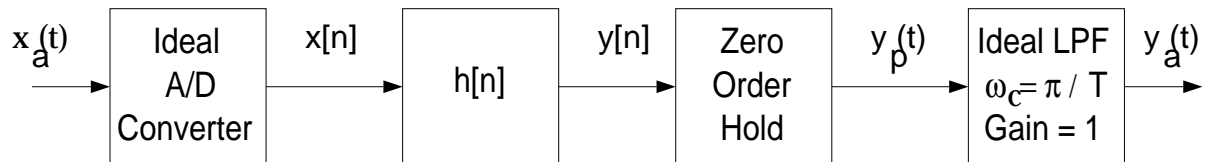
$$y_p(t) = \sum_{k=-\infty}^{\infty} y[k]p(t - kT),$$

and

$$p(t) = \begin{cases} 1, & 0 \leq t \leq T \\ 0, & \text{else.} \end{cases}$$

The signal  $y_p(t)$  is then lowpass filtered with an ideal lowpass filter, with frequency response  $G_a(\Omega)$ ,

$$G_a(\Omega) = \begin{cases} 1, & |\Omega| \leq \frac{\pi}{T} \\ 0, & |\Omega| > \frac{\pi}{T}. \end{cases}$$



- (a) (2 points) Determine  $X_d(\omega)$ , the DTFT of  $x[n]$ , in terms of  $X_a(\Omega)$ .

(b) (2 points) Determine  $Y_d(\omega)$ , the DTFT of  $y[n]$ , in terms of  $X_a(\Omega)$ .

(c) (3 points) Determine  $Y_p(\Omega)$ , the Fourier Transform of  $y_p(t)$ , in terms of  $X_a(\Omega)$ .

(d) (3 points) Determine  $Y_a(\Omega)$ , the Fourier Transform of  $y_a(t)$ , in terms of  $X_a(\Omega)$ . Also find  $H_{eff}(\Omega)$ , the effective frequency response of the entire system that takes  $x_a(t)$  as input and produces  $y_a(t)$  as output.

**Problem 2 (10 points)**

In this problem, you would like to use the window method to design a causal, finite-length impulse response (FIR) filter, which is an approximation to the ideal highpass filter:

$$D(\omega) = \begin{cases} 0, & |\omega| \leq \frac{\pi}{4} \\ 1, & \frac{\pi}{4} < |\omega| < \pi \end{cases}$$

where  $d[n]$  is an infinite-length sequence that is symmetric about  $n = 0$ . Note that your FIR filter will have generalized linear phase.

(a) (2 points) Determine  $d[n]$ , the impulse response of the desired ideal high pass filter.

(b) (3 points) Let  $h[n] = w[n]g[n]$  be the impulse response of your FIR filter, for a rectangular window, with

$$w[n] = \begin{cases} 1, & n = 0, \dots, N - 1 \\ 0, & \text{otherwise.} \end{cases}$$

Determine  $g[n]$ , and also determine  $h[n]$ , the impulse response of the causal FIR approximation of length  $N$ .

- (c) (2 points) Make a rough sketch of what you might expect the magnitude  $|H_d(\omega)|$  to look like. Include any important features. Also make a detailed, accurate plot of the phase,  $\angle [H_d(\omega)]$  labelling any important features.

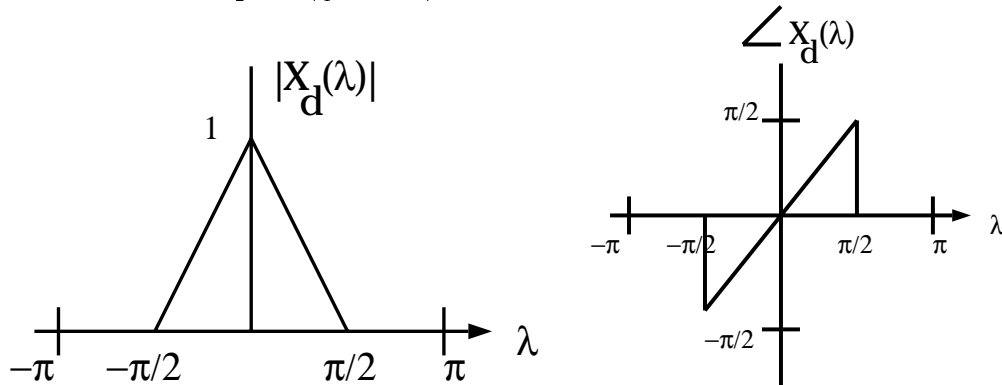
- (d) (3 points) Suppose an analog filter with Laplace transform  $H_L(s)$  has generalized linear phase, i.e.  $H_a(\Omega) = R(\Omega)e^{j(\alpha - M\Omega)}$ . A discrete-time filter is then designed by the following mapping,

$$H(z) = H_L(s) \Big|_s = \frac{1 - z^{-1}}{1 + z^{-1}}$$

Discuss whether or not the discrete-time filter with frequency response  $H_d(\omega)$  will have generalized linear phase?

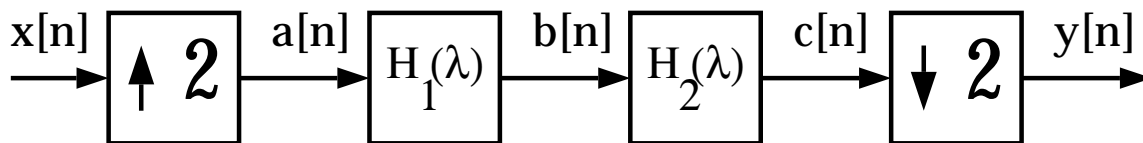
**Problem 3 (10 points)**

Given a signal  $x[n] = [\frac{1}{2}\text{sinc}(\frac{\pi}{4}(n+1))]^2$ , with discrete-time Fourier transform shown below:

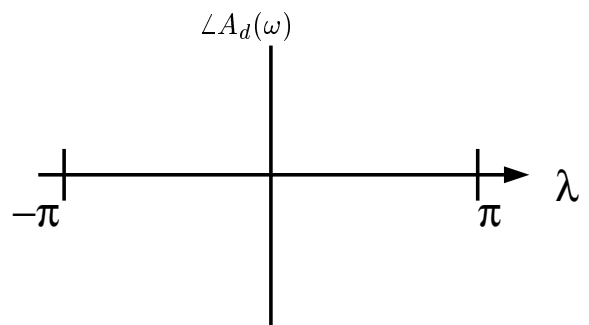
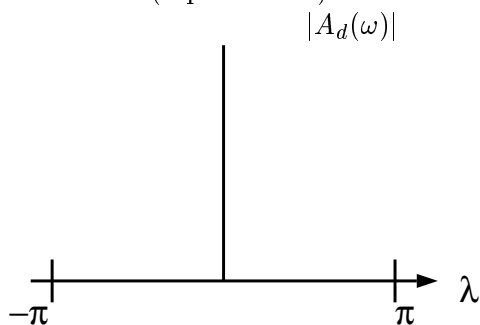


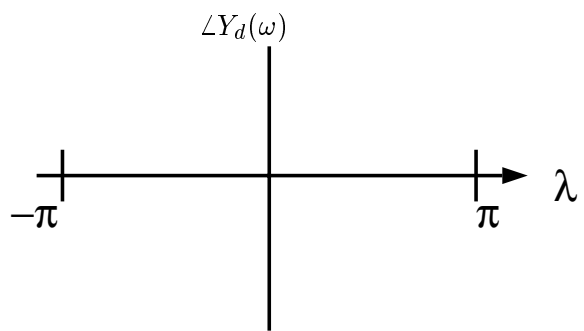
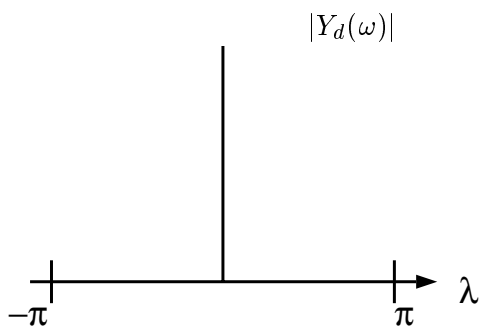
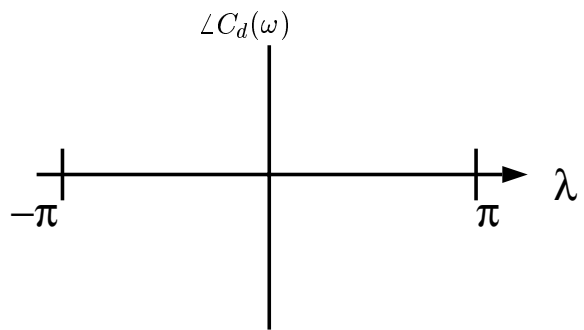
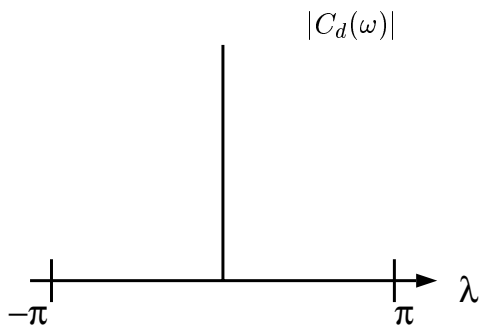
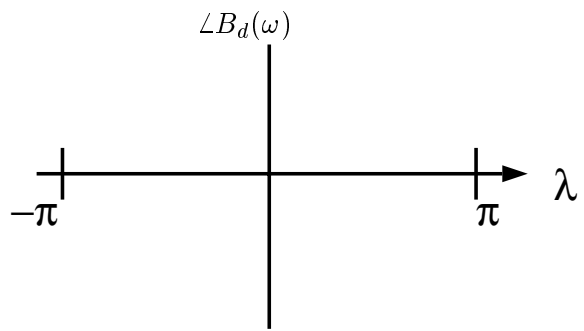
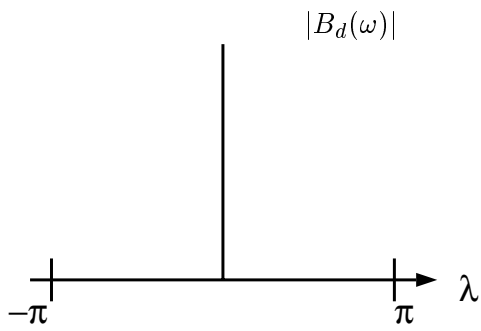
Let  $x[n]$  be the input to the system shown below, where  $H_1(\omega)$ , and  $H_2(\omega)$  are given by

$$H_1(\omega) = \begin{cases} 1, & |\omega| \leq \frac{\pi}{2} \\ 0, & \frac{\pi}{2} < |\omega| < \pi \end{cases} \quad H_2(\omega) = e^{-2j\omega}, \quad |\omega| < \pi$$



- (a) For each of the signals,  $a[n]$ ,  $b[n]$ ,  $c[n]$ , and  $y[n]$ , plot the magnitude and phase of their DTFT below (1 point each).

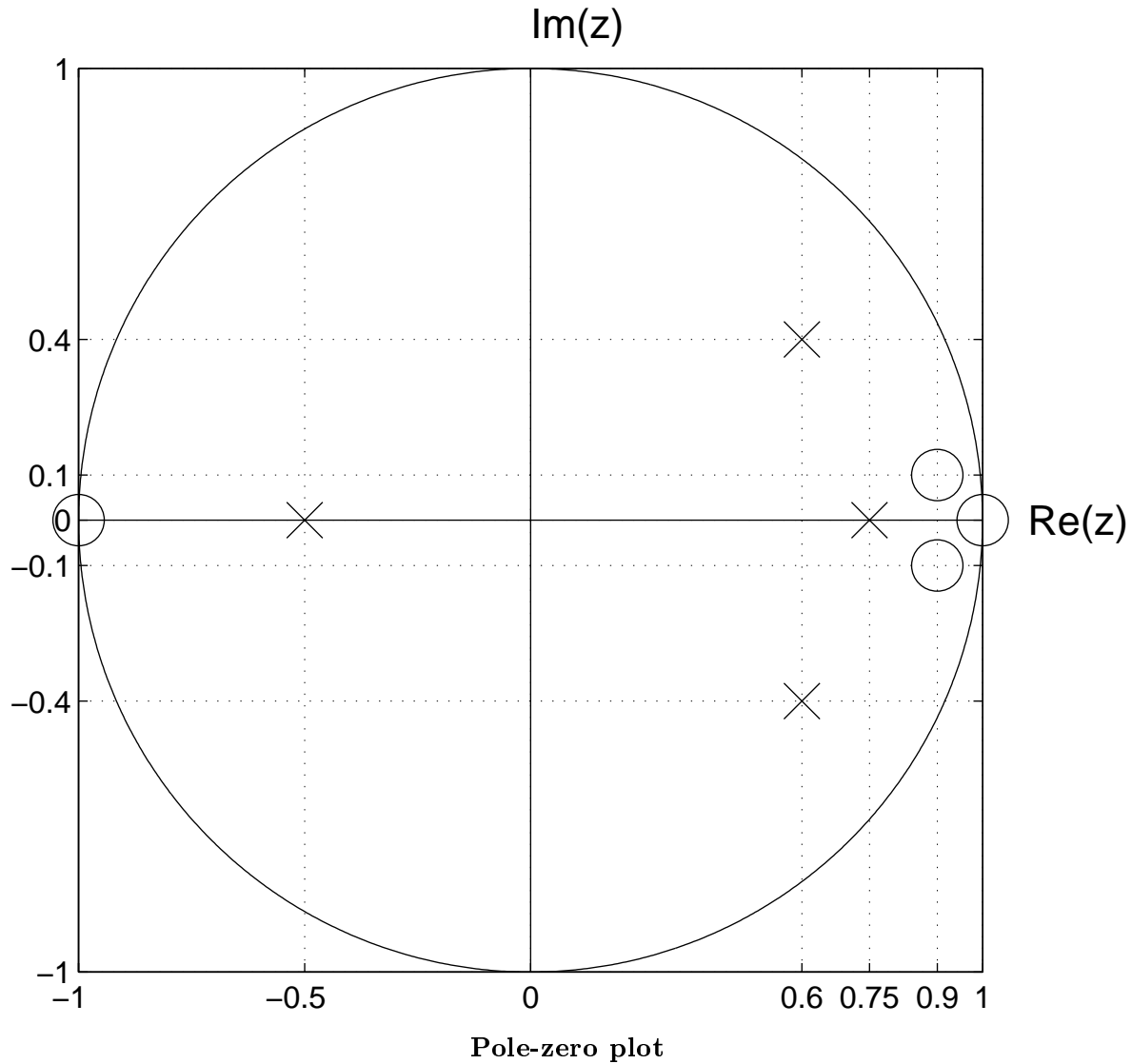




(b) Determine the sequence  $y[n]$ . (2 points)  $y[n] =$  \_\_\_\_\_

**Problem 4 (10 points)**

A causal, stable discrete-time filter with transfer function  $H(z)$  has poles and zeros as shown in the following figure:



and satisfies,  $\lim_{|z| \rightarrow \infty} H(z) = 2$ . On the following page, draw a block diagram for a possible implementation of  $H(z)$  using a cascade of two second-order subsections, where each subsection is in direct form I and has real coefficients.

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Draw your implementation of  $H(z)$  here, and indicate your reasoning.