

UNIVERSITY OF TORONTO
FACULTY OF APPLIED SCIENCE AND ENGINEERING
FINAL EXAMINATION - APRIL 1999
ECE356S - SYSTEM AND SIGNAL ANALYSIS II
Third Year - Programs 5bm(e), 5ce, 5e
Examiner - R.H. Kwong

Students may use one 8.5"×11" aid sheet in preparing their answers. Questions are not of equal weight.

10 points)

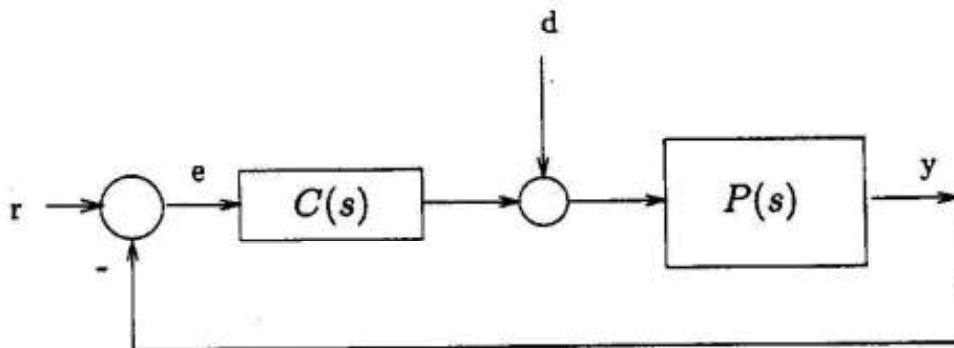
(a) It is known that the matrix

$$A = \begin{bmatrix} 1 & 0 & -42 \\ 0 & 1 & 12 \\ 0 & 0 & -5 \end{bmatrix}$$

has eigenvectors $\begin{bmatrix} 1 \\ 0 \\ 0 \end{bmatrix}$, $\begin{bmatrix} -2 \\ 1 \\ 0 \end{bmatrix}$, $\begin{bmatrix} 7 \\ -2 \\ 1 \end{bmatrix}$. Determine the matrix exponential e^{At} using the diagonalization method.

10 points)

(b) Consider the feedback system



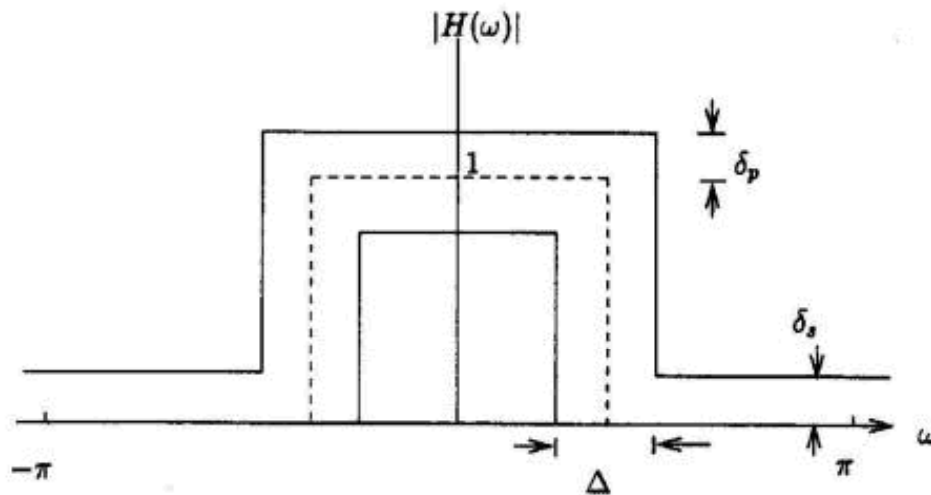
with $r(s) = \frac{1}{s}$, $d(s) = \frac{1}{s^2 + 1}$, and $P(s) = \frac{1}{s - 1}$. The objective is to design a controller $C(s)$ so that the output y tracks the unit step reference input r and that the effects of the disturbance d asymptotically do not appear in the output y . The following controller is proposed:

$$C(s) = \frac{2(s + 1)^2(s + 2)}{s(s^2 + 1)}$$

Will this controller achieve the design specifications? Explain your answer.

10 points)

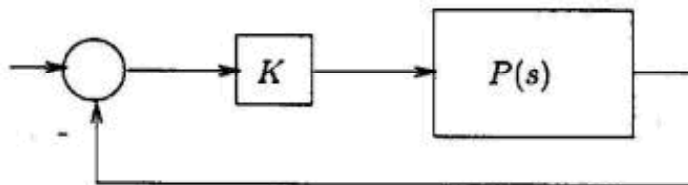
- (c) An FIR low pass digital filter is to be designed using the Kaiser window to satisfy the specifications given in the following figure.



with $\delta_p = 0.00115$, $\delta_s = 0.01$, and $\Delta = 0.1\pi$. Note that the passband ripple specification is different from the stopband ripple specification. Determine the parameters N and α for the Kaiser window needed to satisfy the given specifications.

(25 points)

2. In the following feedback system



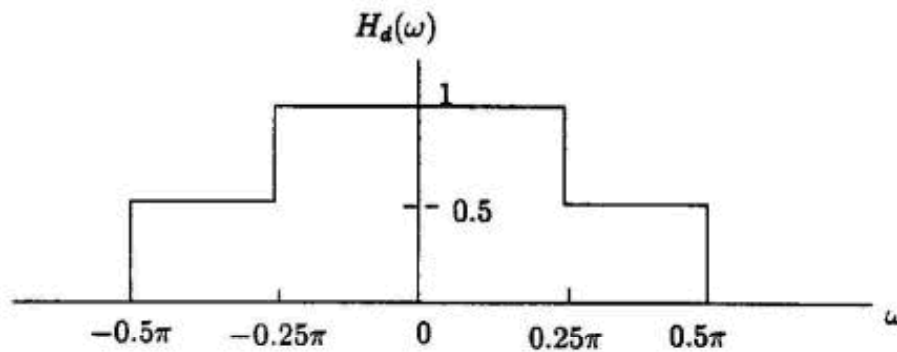
$P(s)$ is given by

$$P(s) = \frac{1 + (\frac{s}{2})}{s^2[1 + (\frac{s}{4})][1 + (\frac{s}{6})]}$$

and K is a positive constant gain. Sketch the Nyquist plot of P . Based on the Nyquist plot, determine the range of values of the gain K for which the closed loop system is stable.

20 points)

3. Suppose we would like to design a 7th order FIR filter ($N = 3$) based on Hamming windows to approximate the following ideal multiband filter



i.e.,

$$H_d(\omega) = \begin{cases} 1, & \text{for } |\omega| \leq 0.25\pi, \\ 0.5, & \text{for } 0.25\pi < |\omega| \leq 0.5\pi, \\ 0 & \text{otherwise} \end{cases}$$

Determine the weights h_k of the Hamming FIR filter

$$y_k = \sum_{j=-3}^3 h_j u_{k-j}$$

(Recall the Hamming window weights are given by $w_k = 0.54 + 0.46 \cos \frac{\pi k}{N}$, $|k| \leq N$.)

(25 points)

4. Design a low-pass Butterworth digital filter based on the bilinear transformation $s = \frac{1 - z^{-1}}{1 + z^{-1}}$ to satisfy the following specifications:

- (i) The cutoff (3-db) frequency ω_c for the digital filter is 0.4688π .
- (ii) For $\omega \geq 0.664\pi$, the attenuation for $|H(e^{j\omega})|$ is at least 15 db.

If there are several factors, you do not need to multiply them out. You are to follow these steps.

- (a) Determine the parameters N and ω_{ac} for the analog Butterworth filter

$$|H_a(\omega)|^2 = \frac{1}{1 + \left(\frac{\omega}{\omega_{ac}}\right)^{2N}}$$

- (b) Determine $H_a(s)$ using the results of (a). You may also, if you wish, use the following table to determine the left half plane roots of $1 + \left(\frac{s}{i}\right)^{2N}$.

N	Roots
2	$(-0.7071 \pm i0.7071)$
3	$-1, -0.5 \pm i0.866$
4	$-0.3827 \pm i0.9239; -0.9239 \pm i0.3827$

Be careful, however, on how to use the table in your calculations.

- (c) Determine $H(z)$.

(Hint: For $s = \frac{1 - z^{-1}}{1 + z^{-1}}$, $\frac{1}{s^2 + as + b} = \frac{(1 + z^{-1})^2}{1 + a + b + (2b - 2)z^{-1} + (1 + b - a)z^{-2}}.$)