

Fifth Semester B.E. Degree Examination, Jan./Feb. 2021

Digital Signal Processing

Max. Ma. Mote: 1. Answer any FIVE full questions, selecting at least TWO full questions

from each part.

2. Use of normalized Chebyshev and Butterworth prototype tables are not allowed.

PART - A

1 a. Find the DFT of a sequence $x(n) = \begin{cases} 1 & \text{for } 0 \le n \le 2 \\ 0 & \text{otherwise} \end{cases}$ for N = 4. Plot magnitude of the

DFT X(k). (06 Marks)

- b. What is the relationship between DFT and DFS? (02 Marks)
- c. If DFT $\{x(n)\} = X(k)$, then show that
 - i) DFT $\{x((-n))_N\} = X((-k))_N$ ii) DFT $\{W_N^{-Ln}x(n)\} = X((k-\ell))_N$ (12 Marks)
- 2 a. Determine the response of an LTI system with $h(n) = \{1, -1, 2\}$ for an input $x(n) = \{1, 0, 1, -2, 1, 2, 3, -1, 0, 2\}$. Use overlap add method with block length N = 4.
 - b. Consider the sequence $x(n) = \{8, 3, 4, 1, -5, -4, -20, 2, -1, 7, 4\}$. Evaluate the following without explicitly computing X(k):

i)
$$\sum_{k=0}^{10} X(k)$$
 ii) $\sum_{k=0}^{10} |X(k)|^2$ (08 Marks)

- 3 a. Derive DIT-FFT algorithm for N = 8 and draw the complete signal flow graph. (12 Marks)
 - b. What is in-place computation? What is the total number of complex additions and multiplications required for N = 512 point, if DFT is computed directly and if FFT is used?

 (04 Marks)
 - c. Compute the 4-point DFT of the sequence $x(n) = \{1, 0, 1, 0\}$ using DIF FFT radix-2-Algorithm. (04 Marks)
- 4 a. Given x(n) = n + 1 and N = 8. Determine X(k) using DIF-FFT algorithm. (12 Marks)
 - b. Write a note on chirp z-transform algorithm. (04 Marks)
 - c. Given $x(n) = \{1, 0, 1, 0\}$, find X(2) using Goertzel algorithm. (04 Marks)

PART - B

- 5 a. Bring out a comparison between Butterworth filter and Chebyshev filter. (04 Marks)
 - b. Design Butterworth filter for following specifications:

$$0.8 \le |H_a(s)| \le 1$$
 for $0 \le f \le 1000 Hz$ (10 Marks)
 $|H_a(s)| \le 0.2$ for $f \ge 5000 Hz$

- c. Derive an expression for order of the Chebyshev filter. (06 Marks)
- 6 a. Design a FIR low pass filter with a desired frequency response

$$H_{d}(w) = \begin{cases} e^{-j3w} & -\frac{3\pi}{4} \le w \le \frac{3\pi}{4} \\ 0 & \frac{3\pi}{4} \le |w| \le \pi \end{cases}$$

Use Hamming window with M = 7. (10 Marks)

b. Design a lowpass FIR filter using frequency sampling technique having cut off frequency of $\frac{\pi}{2}$ rad/sample. The filter should have linear phase and length of 17. (10 Marks)

7 a. Let $H_a(S) = \frac{S+a}{(S+a)^2+b^2}$ be a causal second order analog transfer function. Show that the casual second-order digital function H(z) is obtained from $H_a(S)$ through impulse invariance method is given by

$$H(z) = \frac{1 - e^{-aT} \cos bT z^{-1}}{1 - 2 \cos bT e^{-aT} z^{-1} + e^{-2aT} z^{-2}}$$
(08 Marks)

- b. What are the limitations of Impulse invariance method? (02 Marks)
- c. Design a digital butterworth filter satisfying the following constraints using bilinear transform. Assume T = 1sec.

$$0.9 \le \left| H(e^{jw}) \right| \le 1 \quad 0 \le w \le \frac{\pi}{2};$$

$$\left| H(e^{jw}) \right| \le 0.2 \quad 3\pi/4 \le w \le \pi$$
(10 Marks)

- 8 a. If $H_a(S) = \frac{1}{(S+2)(S+1)}$; find the corresponding H(z) using Impulse invariance method for sampling frequency of 5 samples/sec. (05 Marks)
 - b. Obtain the cascade realization of system function

$$H(z) = 1 + \frac{5}{2}z^{-1} + 2z^{-2} + 2z^{-3}$$
 (05 Marks)

c. Consider the system function

H(z) =
$$\frac{1 + \frac{1}{5}z^{-1}}{\left(1 - \frac{1}{2}z^{-1} + \frac{1}{3}z^{-2}\right)\left(1 + \frac{1}{4}z^{-1}\right)}$$

- i) Realize the system in direct form I
- ii) Realize the system in parallel form using first and second order direct form II sections.
 (10 Marks)