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Sixth Semester B.E. Degree Examination, Jan./Feb. 2021 Digital Signal Processing

Time: 3 hrs.

Max. Marks: 100

Note: I. Answer FIVE full questions, selecting atleast TWO questions from each part.

2. Use of Data sheets may be permitted.

PART - A

1 a. Define DFT and IDFT. State and prove the following properties

i) Periodicity ii) Circular time shift.

(10 Marks)

- b. Compute the 8 point DFT of a sequence $x(n) = (-1)^{n+1}$, $0 \le n \le 7$. Also plot the magnitude and phase plot of DFT. (10 Marks)
- 2 a. Let X(k) be a 11 point DFT of real sequence x(n), where length is 11. The even samples of X(K) are given by X(0) = 2, X(2) = -1 = j3, X(4) = 1 + j4, X(6) = 9 + j3, X(8) = 5, X(10) = 2 j2. Determine the missing odd samples of the DFT. Find x(0) and $\sum x(n)$.

(10 Marks)

- b. Compute the convolution of two sequences using circular array and tabular array. x(n) = (1, 1, 1, 1, -1, -1, -1, -1) and h(n) = (0, 1, 2, 3, 4, 3, 2, 1). (10 Marks)
- 3 a. Using linear convolution find y(n) = x(n) * h(n) for the sequences x(n) = (3, 0, -2, 0, 2, 1, 0, -2, -1, 0) and h(n) = (2, 2, 1). Verify the results by solving the problem using overlap—save method. (14 Marks)
 - b. Tabulate the comparison of complex multiplications and additions for direct computation of DFT versus the FFT algorithm for N = 32,128 and 512. (06 Marks)
- 4 a. Compute the 8 point DFT of sequence $x(n) = \sin \frac{\pi}{2} n$ using DIT FFT algorithm. (10 Marks)
 - b. Find the circular convolution of x(n) = (1, 1, 1, 1) with h(n) = (1, 2, 3, 4) using radix 2 DIF-FFT for DFT's and radix 2 DIT-FFT to find IDFT. (10 Marks)

PART - B

- 5 a. Explain the comparison between:
 - i) Analog and digital filters
 - ii) Butterworth and Chebyshev filter.

(12 Marks)

- b. Obtain the transfer function of IIR digital filter for given Ha(S) using impulse invariance method. Ha(s) = $\frac{0.5(s+4)}{(s+1)(s+2)}$. (08 Marks)
- a. Design an analog Chebyshev filter with the following specifications:

 $A_P = -3db$

 $\Omega_P = 2 \text{ rad/sec}$

for

 $A_S = -20db$

 $\Omega_{\rm S} = 4 \, \rm rad/sec$

Obtain ε , N, H_a(S).

(10 Marks)

b. Design an analog low-pass Butterworth filter by impulse invariance for the following specifications.

 $0.89125 \le |H(\omega)| \le 1$

 $0 \le \omega_{\rm P} \le 0.2\pi$

 $|H(\omega)| \le 0.17783$

for $0.3 \pi \le \omega_S \le \pi$.

(10 Marks)

- 7 a. What are the advantages and disadvantages with the design of FIR filters using window function? (08 Marks)
 - b. The desired frequency response of a low pass filter is given by

$$H_{d}(\omega) = e^{-je\omega} \qquad |\omega| \le \frac{3\pi}{4}$$
$$= 0 \qquad \frac{3\pi}{4} \le |\omega| < \pi$$

Determine the frequency response of the FIR filter if Hamming window is used with N =7.

(12 Marks)

- 8 a. Obtain the direct form I, direct form II realizations for the following systems: $y(n) = 0.75y(n-1) 0.125y(n-2) + 6x(n) + 5x(n-1) + x(n-2). \tag{06 Marks}$
 - b. A discrete time system H(z) is expressed as,

$$H_{d}(z) = \frac{10\left(1 - \frac{1}{2}z^{-1}\right)\left(1 - \frac{2}{3}z^{-1}\right)\left(1 + 2z^{-1}\right)}{\left(1 - \frac{3}{4}z^{-1}\right)\left(1 - \frac{1}{8}z^{-1}\right)\left[1 - \left(\frac{1}{2} + \frac{1}{2}j\right)z^{-1}\right]\left[1 - \left(\frac{1}{2} - \frac{1}{2}j\right)z^{-1}\right]}$$

Realize cascade forms using second order systems.

c. Compare direct form-I and form-II realizations.



(04 Marks)