

# CBCS SCHEME

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17EC52

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

Time: 3 hrs.

Max. Marks: 100

*Note: Answer any FIVE full questions, choosing ONE full question from each module.*

### Module-1

- Explain frequency domain sampling and reconstruction of discrete time signals. (10 Marks)
  - Compute circular convolution of two sequences,  $x_1(n) = \{1, 2, 3, 4\}$  and  $x_2(n) = \{1, -1, 3, 2\}$ , using DFT-IDFT method. (06 Marks)
  - Compute 16-point DFT of the sequence  $x(n) = 8, 0 \leq n \leq 15$ . (04 Marks)

OR

- Compute N-point DFT of the sequence  $x(n) = \sin\left(\frac{2\pi K_0 n}{N}\right), 0 \leq n \leq N-1$ . (08 Marks)
  - Compute DFT of the sequence  $x(n) = \sin\left(\frac{3\pi n}{4}\right) + \cos\left(\frac{\pi n}{4}\right), 0 \leq n \leq 3$ , using linearity property of DFT. (06 Marks)
  - Derive the relationship between DFT and DTFS coefficients. (06 Marks)

### Module-2

- The 4-point DFT of a length-4 sequence  $x(n)$  is given by  $X(k) = \{8, -1+j, -2, -1-j\}$ . Obtain  $y(k)$ , the 4-point DFT of the sequence  $y(n) = e^{-j\pi n} x((n-1))_4$ . (05 Marks)
  - Given a sequence  $x(n) = \{1, -1, 2, -2\}$ , determine DFT  $\{DFT\{DFT\{DFT\{x(n)\}\}\}$ , using complex conjugate properties of DFT. (07 Marks)
  - Determine the filter output  $y(n)$ , whose impulse response  $h(n) = \{1, -1, 2\}$  and input  $x(n) = \{1, 4, 3, 2, 1, -1, 2, 1, 5, 3, 2, 4\}$ , using overlap-save method. Consider 8-point circular convolution approach. (08 Marks)

OR

- The 4-point DFT of a sequence  $x(n)$  is given by  $x(k) = \{16, -4+j4, -4, -4-j4\}$ . Determine the energy of  $x(n)$  using Parseval's theorem. (04 Marks)
  - The IDFT  $\{x(k)\}$  is given by  $x(n) = \{1, 2, 3, 4\}$ . Determine IDFT of the following sequences: i)  $x(4-k)$  ii)  $j^k x(k)$  iii)  $\text{Re}\{x(k)\}$  iv)  $\text{Im}\{x(k)\}$  (10 Marks)
  - Discuss the need of FFT algorithms for computation of DFT. (06 Marks)

### Module-3

- Compute 8-point DFT of the sequence  $x(n) = \{0.707, 0, -0.707, -1, -0.707, 0, 0.707, 1\}$  using DIT-FFT algorithm. (08 Marks)
  - Starting from the expression of Z-transform of an N-point sequence  $x(n)$ , derive chirp z-transform algorithm. (08 Marks)
  - Mention the similarities and differences between DIT-FFT and DIF-FFT algorithm. (04 Marks)

Important Note : 1. On completing your answers, compulsorily draw diagonal cross lines on the remaining blank pages.  
2. Any revealing of identification, appeal to evaluator and /or equations written eg. 42+8 = 50, will be treated as malpractice.

OR

- 6 a. Develop the radix-2 DIF-FFT algorithm for  $N = 8$  and draw the signal flow graph. (10 Marks)  
 b. Given  $x(n) = \{1, 2, 3, -1\}$ , obtain  $X(1)$  using Goertzel algorithm and also explain Goertzel Algorithm. (10 Marks)

Module-4

- 7 a. Obtain a parallel realization for the transfer function  $H(z)$  given below:

$$H(z) = \frac{8z^3 - 4z^2 + 11z - 2}{\left(z - \frac{1}{4}\right)\left(z^2 - z + \frac{1}{2}\right)} \quad (06 \text{ Marks})$$

- b. Derive an expression for order and cut-off frequency of low-pass Butterworth filter. (08 Marks)

- c. Transform the analog filter,

$$H_a(s) = \frac{s+1}{s^2+5s+6}$$

into digital filter,  $H(z)$  using impulse invariant transformation. Consider  $T = 0.1$  sec. (06 Marks)

OR

- 8 a. Design a digital filter  $H(z)$  that when used in A/D -  $H(z)$  - D/A structure gives an equivalent analog filter with the following specifications: Passband attenuation  $\leq 3.01$ dB, Passband edge frequency = 500Hz, Stopband attenuation  $\geq 15$ dB, Stopband edge frequency = 750Hz and sampling rate = 2kHz. The filter is to be designed by performing bilinear transformation on Butterworth analog filter. (12 Marks)

- b. A linear time-invariant digital IIR filter is specified by the transfer function,

$$H(z) = \frac{(z^2 - 1)(z^2 - 2z)}{\left(z^2 + \frac{1}{16}\right)\left(z^2 - z + \frac{1}{2}\right)}$$

Obtain direct form-I and direct form-II realizations of the system. (08 Marks)

Module-5

- 9 a. A filter is to be designed with the following desired frequency response:

$$H_d(\omega) = \begin{cases} 0, & |\omega| < \pi/4 \\ e^{-j2\omega}, & \pi/4 < |\omega| < \pi \end{cases}$$

Find the frequency response of the FIR filter designed using rectangular window. (10 Marks)

- b. Given the FIR filter with the following difference equation:

$$y(n) = x(n) + 3.1x(n-1) + 5.5x(n-2) + 4.2x(n-3) + 2.3x(n-4)$$

Sketch the lattice realization of the filter. (10 Marks)

OR

- 10 a. The frequency response of an ideal band pass filter is given by;

$$H_d(\omega) = \begin{cases} e^{-j3\omega}, & 1 < |\omega| < 2 \\ 0, & |\omega| < 1 \text{ or } 2 < |\omega| < \pi \end{cases}$$

Design an FIR bandpass filter which approximates the above filter, using Hamming window. (10 Marks)

- b. Realize the linear-phase FIR filter having the following impulse response:

$$h(n) = \delta(n) + \frac{1}{4}\delta(n-1) - \frac{1}{8}\delta(n-2) + \frac{1}{4}\delta(n-3) + \delta(n-4) \quad (05 \text{ Marks})$$

- c. Realize an FIR filter with impulse response  $h(n)$  given by,  $h(n) = \left(\frac{1}{2}\right)^n [u(n) - u(n-4)]$ ,

using direct form-I. (05 Marks)