

# CBCS SCHEME



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15EE63

Sixth Semester B.E. Degree Examination, Dec.2019/Jan.2020

## Digital Signal Processing

Time: 3 hrs.

Max. Marks: 80

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

### Module-1

- a. Find 4 point DFT of  $x(n) = \{1, -2, 3, 4\}$  and plot magnitude and phase response. (06 Marks)  
b. If  $x_1(n) = \{2, 3, 1, 1\}$  and  $x_2(n) = \{1, 3, 5, 3\}$ , find  $x_3(n) = x_1(n) \otimes x_2(n)$  use matrix method (06 Marks)  
c. Prove the time reversal property of DFT. (04 Marks)

OR

- a. Perform circular convolution of  $x_1(n) = \{2, 1, 2, 1\}$  and  $x_2(n) = \{1, 2, 3, 4\}$  using circular shift method. (05 Marks)  
b. Find linear convolution using DFT for the given sequence  $x(n) = \{1, 2, 3\}$  and  $h(n) = \{1, 2, 2, 1\}$ . (06 Marks)  
c. Find the IDFT of the given sequence  $x(k) = \{3, 2 + j, 1, 2 - j\}$ . (05 Marks)

### Module-2

- a. Find the 8-point DFT of sequence  $x(n) = \{1, 1, 1, 1, 0, 0, 0, 0\}$  using DIT FFT radix 2 algorithm. Draw signal graph. (08 Marks)  
b. Develop a Decimation in Frequency FFT algorithm for  $N = 8$ . Draw signal flow graph. (08 Marks)

OR

- a. Develop a decimation in time algorithm FFT of  $N = 8$  draw signal flow graph. (08 Marks)  
b. Calculate 8-point DFT of sequence  $x(n) = \{1, -1, -1, -1, 1, 1, 1, -1\}$ , using DIF - FFT radix -2 algorithm. (08 Marks)

### Module-3

- a. Design an analog Chebyshev with following specification.  
Passband : 1db for  $0 \leq \Omega \leq 10$  rad/sec  
Stopband attenuation : -60 db for  $\Omega \geq 50$  rad/sec. (10 Marks)  
b. The system function of an analog filter is given as  $H_a(s) = \frac{1}{(s+1)(s+2)}$ . Obtain  $H(z)$  using impulse invariant method take sampling frequency as 5 samples/sec. (06 Marks)

OR

- a. Design a low pass Butterworth filter using bilinear transformation method to meet the following specification take  $T = 2$ sec  
Passband ripple  $\leq 1.25$ dB  
Passband edge = 200 Hz  
Stopband attenuation  $\geq 15$ dB  
Stopband edge = 400Hz  
Sampling frequency = 2KHz (10 Marks)  
b. Prove the following transformation relation for impulse invariant transform.

$$\frac{s+a}{(s+a)^2 + b^2} = \frac{1 - e^{-aT}(\cos bT)z^{-1}}{1 - 2e^{-aT}(\cos bT)z^{-1} + e^{-2aT}z^{-2}} \quad (06 \text{ Marks})$$

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.

**Module-4**

- 7 a. Compare bilinear transformation with impulse invariance transformation. (04 Marks)  
 b. Write a note on frequency warping. (06 Marks)  
 c. Determine Direct form – I and II for 2<sup>nd</sup> order filter given by  
 $y(n) = 2b \cos w_0 y(n-1) - b^2 y(n-2) + x(n) - b \cos w_0 x(n-1)$  (06 Marks)

**OR**

- 8 a. Obtain the Cascade form realization for given system.  

$$H(z) = \frac{(z-1)(z-2)(z+1)z}{\left(z - \frac{1}{2} - \frac{1}{2}j\right)\left(z - \frac{1}{2} + \frac{1}{2}j\right)\left(z - \frac{1}{4}j\right)\left(z + \frac{1}{4}j\right)}$$
 (08 Marks)  
 b. Design a second order lowpass digital Butterworth filter with cutoff frequency 1KHz and sampling frequency of  $10^4$  samples/sec by linear transformation. (08 Marks)

**Module-5**

- 9 a. Given the FIR filter with following difference equation  
 $y(n) = x(n) + \frac{2 \cdot x}{5}(n-1) + \frac{3}{4}x(n-2) + \frac{1}{3}x(n-3)$ . Draw direct Form – I and lattice structure. (08 Marks)  
 b. Using frequency sampling method, design a band pass filter with following specification determine the filter coefficient for  $N = 7$ , sampling frequency  $F = 8000\text{Hz}$ , cutoff frequency  $f_{c_1} = 1000\text{Hz}$ ,  $f_{c_2} = 3000\text{Hz}$  (08 Marks)

**OR**

- 10 a. Realise the following system function in cascade form  
 $H(z) = 1 + \frac{3}{4}z^{-1} + \frac{17}{8}z^{-2} + \frac{3}{4}z^{-3} + z^{-4}$  in direct form I and cascade form. (08 Marks)  
 b. Design the symmetric FIR lowpass filter whose desired frequency response is given as  

$$H_d(\omega) = \begin{cases} e^{-j\omega z}, & \text{for } |\omega| \leq \omega_c \\ 0, & \text{otherwise} \end{cases}$$
  
 The length of filter should be 7 and  $\omega_c = 1$  rad/sample use rectangular window. (08 Marks)

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