

CBCS SCHEME

17EC52

## Fifth Semester B.E. Degree Examination, Aug./Sept.2020 **Digital Signal Processing**

Time: 3 hrs.

Max. Marks: 100

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

Module-1

- Prove that the uniform sampling of Discrete Time Fourier Transform of a sequence, x(n) 1 results in N - point DFT.
  - Evaluate the N point DFT of a sequence  $x(n) = 1 + \cos^2\left(\frac{2\pi n}{N}\right)$ ;  $0 \le n \le N 1$ . (07 Marks)
  - Derive the relationship of N point DFT with Z transform.

(06 Marks)

OR

- Define the DFT and IDFT of a sequence. Show that N point DFT and IDFT are periodic 2 with period 'N'. (07 Marks)
  - Let x(n) be a finite length sequence with its DFT  $x(k) = \{1, 4i, 0, -4i\}$ . Find the DFTs of
    - $x_1(n) = e^{\frac{j\pi n}{2}} \cdot x(n)$
- ii)  $x_2(n) = \cos$

Keep answer in terms of x(k).

(07 Marks)

Compute the IDFT of a sequence  $x(k) = \{24, -2j, 0, 2j\}$ .

(06 Marks)

- State and prove i) Circular time shift property ii) Circular convolution property of DFTs. 3 (08 Marks)
  - Define Twiddle factor. Prove the following properties of Twiddle factor:
    - Periodicity property
- ii) Symmetry property.

(04 Marks)

Find the output y(n) of a filter whose impulse response  $h(n) = \{1, 2, 3, 4\}$  for an input x(n) $x(n) = \{1, 2, 1, -1, 3, 0, 5, 6, 2, -2, -5, -6, 7, 1, 2, 0, 1\}$ . Using overlap – add method. (08 Marks) Use 6 – point circular convolution. CMRIT LIBRARY

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In direct computation of N – point DFT, how many i) Complex additions ii) Complex Trigonometric functions are required to calculate. Also explain the multiplications iii) need of FFT algorithms. (06 Marks)

- b. Given  $x_1(n) = \{1, 2, 3, 4\}$  and  $x_2(n) = \{1, 2, 2\}$ . Compute circular convolution  $x_3(n)$  of  $x_1(n)$ with  $x_2(n)$  using Concentric Circle method. (07 Marks)
- Compute convolution of  $x(n) = \{1, 2, 3\}$  with  $h(n) = \{4, 5\}$  using DFT and IDFT method.

(07 Marks)

Module-3

- Find 8 point DFT of a sequence  $x(n) = \{1, 2, 3, 4, 4, 3, 2, 1\}$  using Radix 2 DIT FFT 5 algorithm.
  - b. What is Geortzel Algorithm? Obtain Direct form II structure for the Geortzel filter. (10 Marks)

## OR

- 6 a. Develop Radix -2 DIF FFT algorithm and draw complete signal flow graph for N=8.

  (10 Marks)
  - b. Compute the 8- point IDFT of a sequence.  $X(k) = \{7, -0.707 j0.707, -j, 0.707 j0.707, 1, 0.707 + j0.707, j, -0.707 + j0.707\}$  using Radix 2 DIF FFT algorithm. (10 Marks)

Module-4

- 7 a. Derive an expression for the order of analog Butterworth prototype low pass filter. (08 Marks)
  - b. Design a digital Butterworth filter using Bilinear transformation method to meet following:
    - i) Stopband attenuation ≤ 1.25 dB at passband edge frequency of 200Hz and
    - ii) Stopband attenuation ≥ 15dB at stopband edge frequency of 400Hz. Take sampling frequency of 2KHz. (12 Marks)

## OR

8 a. An analog third order Butterworth low-pass filter has the transfer function

 $H_9(s) = \frac{1}{(s+1)(s^2+s+1)}$ . Design the corresponding digital filter using impulse invariance

method. (08 Marks)

b. Obtain direct form – I , direct form – II , Cascade form and Parallel form realization of the system defined by

$$H(z) = \frac{(z-1)(z-3)(z^2+5z+6)}{(z^2+6z+5)(z^2-6z+8)}.$$
 (12 Marks)

## Module-5

9 a. Design a linear – phase high pass FIR filter using Hamming window for the following desired frequency response.

 $H_{d}(e^{jw}) = \begin{cases} 0 & ; & |w| < \frac{\pi}{4} \\ e^{-j2w} & ; & \frac{\pi}{4} \le |w| \le \pi \end{cases}$  (08 Marks)

b. An FIR filter is defined by difference equation;

 $y(n) = 2.x(n) + \frac{4}{5}x(n-1) + \frac{3}{2}x(n-2) + \frac{2}{3}x(n-3)$ . Find lattice coefficients. Also draw direct

form and lattice form.
c. Compare FIR filter with IIR filter.

(08 Marks) (04 Marks)

- OR
- 10 a. Design a linear phase FIR filter using rectangular window for the following desired frequency response

 $H_{d}(e^{jw}) = \begin{cases} e^{-j2w} & ; & |w| < \frac{\pi}{4} \\ 0 & ; & \frac{\pi}{4} \le |w| \le \pi \end{cases}$  (08 Marks)

b. Realize the FIR filter whose transfer function is given by  $H(z) = 1 + \frac{3}{4} Z^{-1} + \frac{17}{8} Z^{-2} + \frac{3}{4} Z^{-3} + Z^{4}$ . Using Direct form – I and Linear phase form.

(08 Marks)

c. Explain Gibbs phenomenon. Also mention methods to minimize it.

(04 Marks)