USN

Sixth Semester B.E. Degree Examination, Dec.2016/Jan.2017 **Digital Signal Processing**

Time: 3 hrs. Max. Marks:100

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

PART - A

- 1 a. If X(k) is N point DFT of N-length sequence x(n), and if $x_1(n)$ is DFT of X(k), then determine $x_1(n)$ interms of x(n).
 - b. Compute 8 point DFT of the sequence $x(n) = \{1, 2, 2, 1, 2, 2\}$ and verify conjugate symmetry about k = N/2. (10 Marks)
 - c. If X(k) represent 6-point DFT of sequence. $X(n) = \{2, -1, 3, 4, 0, 5\}$, then find y(n) of same length as x(n) such that its DFT $Y(k) = W_3^{2k}X(k)$. (05 Marks)
- 2 a. Using Stockham's method find circular convolution of the sequences : $g(n) = \delta(n) + 2\delta(n-1) + 3\delta(n-2) + 4\delta(n-3)$ and h(n) = n for $0 \le n \le 3$.

b. Obtain output of the system having impulse response $h(n) = \cos(\frac{2\pi n}{N})$ and

input $x(n) = \sin(\frac{2\pi n}{N})$, through N – point circular convolution.

(06 Marks)

(07 Marks)

- c. Use sectional convolution approach to find the response of filter having impulse response $h(n) = \{1, 2\}$ and input $x(n) = \{1, 2, -1, 2, 3, -2, -3, -1, 1, 1, 2, -1\}$. Use 5-point circular convolution use overlap and add method. (07 Marks)
- 3 a. Develop DIF FFT algorithm for N = 8 from basic principles of decomposition of radix-2. (10 Marks)
- 4 a. The first five points of DFT of a sequence are given as {7, -0.707-j0.707, -j, 0.707-j0.707, 1}. Obtain the corresponding time domain sequence of length-8 using DIF FFT algorithm.

 (10 Marks)
 - b. Develop a N-composite DIT FFT algorithm for evaluating 9 point DFT. (10 Marks)

PART - B

5 a. A lowpass Butterworth filter has to meet the following specifications:

Passband gain, $K_p = -1 dB$ at $\Omega_p = 4$ rad/sec

Stopband attenuation greater than or equal to 20 dB at $\Omega_S = 8$ rad/sec.

Determine the transfer function $H_a(s)$ of the lowest order Butterworth filter to meet the above specifications. (10 Marks)

b. Design a Chebyshev – I filter to meet the following specifications:

Passband ripple

: ≤ 2dB

Passband edge

: 1 rad/sec

Stopband attenuation Stopband edge : ≥ 20 dB : 1.3 rad/sec.

(10 Marks)

6 a. Using impulse invariant transformation, design a digital Chebyshev I filter that satisfies the following constraints. $0.8 \le |H(\omega)| \le 1$, $0 \le \omega \le 0.2\pi$

$$|H(\omega)| \le 0.2$$
, $0.6\pi \le \omega \le \pi$.

(12 Marks)

- b. Define the following windows along with their impulse response:
 - i) Rectangular window
 - ii) Hamming window
 - iii) Hanning window.

(08 Marks)

7 a. The desired frequency response of a lowpass FIR filter is given by:

$$H_{d}(\omega) = \begin{cases} e^{-j3\omega}, & |\omega| < \frac{3\pi}{4} \\ 0, & \frac{3\pi}{4} < |\omega| < \pi \end{cases}$$

Determine the frequency response of the filter using Hamming window for N = 7. (10 Marks)

b. Determine the filter coefficients h(n) obtained by sampling $H_d(\omega)$ given by :

$$H_{d}(\omega) = \begin{cases} e^{-j3\omega}, & 0 < \omega \le \frac{\pi}{2} \\ 0, & \frac{\pi}{2} < \omega < \pi \end{cases}$$

Also obtain frequency response taking N = 7.

(10 Marks)

8 a. For a LTI system described by following input-output relation:

$$2y(n) - y(n-2) - 4y(n-3) = 3x(n-2)$$

Realize the system in following forms:

- i) Direct form I
- ii) Direct form II transposed realization.

(10 Marks)

b. Obtain cascade realization for the system function given below:

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

c. Compare direct from - I and II realizations.

(04 Marks)

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