

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

Time: 3 hrs.

Max. Marks:100

Note: Answer FIVE full questions, selecting at least TWO questions from each per

PART - A

Define DFT and IDFT of a signal. Establish relation between DFT and Z-transform. 1

(06 Marks)

Find the IDFT of x(k) = (24, -2j, 0, +2j)b.

(06 Marks)

Find the 8-point DFT of the sequence $x(n) = \{1, 1, 1, 0\}$ c.

(08 Marks)

State and prove the circular (i) Time-shift and (ii) Frequency – shift properties of an N-point 2 a. sequence. (06 Marks)

b. Find the 4-point circular convolution of the sequences

(04 Marks)

 $x_1(n) = (1, 2, 3, 1)$ and $x_2(n) = (4, 3, 2, 2)$.

c. Let x(k) be a 14-point DFT of length – 14 real sequence x(n). The first 8-samples of x(k) are given by x(0) = 12, x(1) = -1+3j, x(2) = 3+4j, x(3) = 1-5j, x(5) = 6+3j, x(6) = -2-3j, x(7) = 10. Find the remaining samples of x(k). Also evaluate the following:

ii) x(7) iii) $\sum_{n=0}^{13} x(n)$ iv) $\sum_{n=0}^{13} |x(n)|^2$

(10 Marks)

In the direct computation of N-point DFT of x(n), how many

i) Complex additions

- ii) Complex multiplications
- iii) Real multiplication
- iv) Real additions and

v) Trigonometric functions, evaluations are required?

(10 Marks)

b. Find the output y(n) of a filter whose impulse response $h(n) = \{1, 2, 3, 4\}$ and the input signal to the filter is $x(n) = \{1, 2, 1, -1, 3, 0, 5, 6, 2, -2, -5, -6, 7, 1, 2, 0, 1\}$ using overlap add method with 6-point circular convolution. (10 Marks)

What is chirp-z-transform? Mention its applications. a.

(04 Marks)

b. Given $x(n) = \{1, 0, 1, 0\}$, find x(2) using Goertzel algorithm.

(06 Marks)

Determine 8-point DFT of a signal x(n) using, Radix - 2 DIF-FFT algorithm, draw the signal flow graph. $x(n) = \{0, 0.707, 1, 0.707, 0, -0.707, -1, -0.707\}$ (10 Marks)

PART – B

For Analog Butterworth filter, derive an expression for order, cut off frequency for design of 5 low pass filter.

(10 Marks)

(10 Marks)

b. Design Butterworth filter for following specifications: $0.8 \le \text{Ha}(s) \le 1$ for $0 \le F \le 1\text{KHz}$ and $|\text{Ha}(s)| \le 0.2$ for $F \ge 5\text{KHz}$

1 of 2

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



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- 6 a. Realize an FIR filter given by $h(n) = \left(\frac{1}{2}\right)^n \left[u(n) u(n-4)\right]$ using direct form I. (06 Marks)
 - b. Obtain the direct form I, direct form II, cascade and parallel form realization for the following system.
 Y(n) = 0.75 y(n-1) 0.125y (n-2) + 6x(n) + 7x(n-1)+ x(n-2).
- 7 a. Write equations of any four different windows used in design of FIR filters. (08 Marks)
 - b. Design the symmetric FIR, low pass filter whose desired frequency response is given as, $H_d(w) = \begin{cases} e^{-jw\tau}, & \text{for } |w| \leq w_c \\ 0, & \text{otherwise} \end{cases}$

The length of the filter should be 7 and $w_c = 1$ radian/sample. Use rectangular window.

(12 Marks)

- 8 a. Explain how analog filter is mapped on to a digital filter using impulse invariant method.

 (08 Marks)
 - b. Design a digital low pass filter to satisfy the following pass band ripple $1 \le H(j\Omega) \le 0$, for $0 \le \Omega \le 1404\pi$ rad/sec and stop band attenuation $|H(\Omega)| > 60 dB$ for $\Omega \ge 8268\pi$ rad/sec. sampling interval $T_s = \frac{1}{10^{-4}}$ sec. Use BLT for designing. (12 Marks)

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