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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 part.**

PART – A

- 1 a. Define DFT and IDFT of a signal. Establish relation between DFT and Z-transform. (06 Marks)
 b. Find the IDFT of $x(k) = (24, -2j, 0, +2j)$ (06 Marks)
 c. Find the 8-point DFT of the sequence $x(n) = \{1, 1, 1, 0\}$. (08 Marks)
- 2 a. State and prove the circular (i) Time-shift and (ii) Frequency – shift properties of an N-point sequence. (06 Marks)
 b. Find the 4-point circular convolution of the sequences $x_1(n) = (1, 2, 3, 1)$ and $x_2(n) = (4, 3, 2, 2)$. (04 Marks)
 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 :
 i) $x(0)$ ii) $x(7)$ iii) $\sum_{n=0}^{13} x(n)$ iv) $\sum_{n=0}^{13} |x(n)|^2$ (10 Marks)
- 3 a. 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)
- 4 a. What is chirp-z-transform? Mention its applications. (04 Marks)
 b. Given $x(n) = \{1, 0, 1, 0\}$, find $x(2)$ using Goertzel algorithm. (06 Marks)
 c. 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

- 5 a. For Analog Butterworth filter, derive an expression for order, cut off frequency for design of low pass filter. (10 Marks)
 b. Design Butterworth filter for following specifications : (10 Marks)
 $0.8 \leq |H_a(s)| \leq 1$ for $0 \leq F \leq 1\text{KHz}$ and $|H_a(s)| \leq 0.2$ for $F \geq 5\text{KHz}$

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.

- 6 a. Realize an FIR filter given by $h(n) = \left(\frac{1}{2}\right)^n [u(n) - u(n-4)]$ using direct form - I. (06 Marks)
- b. Obtain the direct form - I, direct form - II, cascade and parallel form realization for the following system.
 $Y(n) = 0.75 y(n-1) - 0.125y(n-2) + 6x(n) + 7x(n-1) + x(n-2)$. (14 Marks)
- 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 $1 \leq H(j\Omega) \leq 0$, for $0 \leq \Omega \leq 1404\pi$ rad/sec and stop band attenuation $|H(\Omega)| > 60\text{dB}$ for $\Omega \geq 8268\pi$ rad/sec. sampling interval $T_s = \frac{1}{10^4}$ sec. Use BLT for designing. (12 Marks)

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