

**GUJARAT TECHNOLOGICAL UNIVERSITY**  
**M. E. - SEMESTER – I • EXAMINATION – WINTER • 2013**

**Subject code: 714204N****Date: 06-01-2014****Subject Name: Advance DSP System & Architecture****Time: 10.30 am – 01.00 pm****Total Marks: 70****Instructions:**

1. Attempt all questions.
2. Make suitable assumptions wherever necessary.
3. Figures to the right indicate full mark.

- Q.1** (a) List properties of LTI Systems and describe its conditions. **07**  
 (b) Define: **07**  
 1) Nyquist rate 2) ROC of Z –transform 3) Parseval’s relation for Fourier transform 4) Gibbs Phenomenon 5) All pass filter 6) minimum phase system 7) FFT
- Q.2** (a) Discuss frequency response of LTI system. **07**  
 (b) Sketch the magnitude and phase response of following LTI system. **07**  
 $H(n) = \{1/2, 1/2\}$
- OR**
- (b) Compare various windows used for FIR filter design. **07**
- Q.3** (a) List Properties of Z-Transform and prove two of them. **07**  
 (b) Find Z- Transform of : a)  $h[n] = \{1/3, 1/3, 1/3\}$  b)  $h\{[n] = \delta[n+k] , k>0$  **07**
- OR**
- Q.3** (a) List Properties of DFT and explain two of them in brief. **07**  
 (b) Find IDFT of  $\{2, 1+j, 0, 1-j\}$ . **07**
- Q.4** (a) A filter is described by  $y[n]-(3/4) y[n-1]+(1/8) y[n-2] = x[n] + (1/2) x[n-1]$  **07**  
 Draw a) Direct form I, b) Direct form II realization.
- (b) Explain radix 4 decimation in time FFT algorithm. **07**
- OR**
- Q.4** (a) Perform circular convolution of  $X_1[n] = \{1, 3, 5, 3\}$  and  $X_2[n] = \{2, 3, 1, 1\}$  **07**  
 (b) For a 9 tap FIR filter having zeros at  $z_1 = -0.5, z_2 = 0.3 + j0.5, z_3 = -(1/2) + j\sqrt{3}/2$ . Determine the locations of other zeros and find out the transfer function. **07**
- Q.5** (a) Explain interrupts of ‘C6X processor. **07**  
 (b) Explain the code generation steps in code composer studio. **07**
- OR**
- Q.5** (a) Compare the following **07**  
 1) DSP Processor and Microprocessor  
 2) Von Nu Neumann Architecture and Harvard Architecture
- (b) Write in Brief: (Any two) **07**  
 1) Decimation in time FFT algorithms  
 2) Impulse invariance method for IIR filter Design  
 3) Application of DSP

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