



**VII Semester B.E. (Electrical & Electronics) Degree Examination,
January 2013
(2K6 Scheme)
EE 704 : DIGITAL SIGNAL PROCESSING**

Time : 3 Hours

Max. Marks : 100

Instruction : Answer *five* full questions.

1. a) Prove that sampling of DTFT results in N-point DFT from which the sequence as well as DTFT can be reconstructed. **10**
b) Consider the sequence $x(n) = 4\delta(n) + 3\delta(n-1) + 2\delta(n-2) + \delta(n-3)$. Let $X(K)$ be the 6-point DFT of $x(n)$. Find the finite length sequence $y(n)$ that has 6 point DFT $Y(K) = W_6^{4K}X(K)$. **6**
c) Derive the relationship between N point DFT and Z Transform. **4**
2. a) Let $x(n)$ be a length 'N' real sequence with N-point DFT $X(K)$. Prove that $X(N-K) = X^*(K)$ and $X(0)$ is real. **6**
b) Determine the N- point circular convolution of $x_1(n)$ and $x_2(n)$ defined by
$$x_1(n) = \cos \frac{2\pi n}{N} \quad x_2(n) = \sin \frac{2\pi n}{N}.$$
 6
c) Evaluate the following function for a given 12 point seq.
$$x(n) = [8, 4, 7, -1, 2, 0, -2, -4, -5, 1, 4, 3]$$
 function $\sum_{k=0}^{11} e^{-j\frac{4\pi k}{6}} X(K)$. **8**
3. a) Using DIF FFT, find the 8-point DFT of the sequence $x(n) = [1, 1, 1, 1, 1, 1, 1, 1]$. **10**
b) Discuss the computational benefits of DIT algorithm over direct computation of DFT. **5**
c) Explain with neat contour diagrams the importance and application of Chirp-z transform. **5**



4. a) Realise the linear phase FIR filter having the following impulse response

$$h(n) = \{1, \frac{1}{4}, \frac{1}{8}, \frac{1}{4}, 1\}. \quad 6$$

- b) Obtain a direct form II and cascade realisation for the system function

$$H(z) = \frac{(1+z^{-1})}{\left(1-\frac{1}{4}z^{-1}\right)\left(1-z^{-1}+\frac{1}{2}z^{-2}\right)} \quad 6$$

- c) Determine the coefficients K_m of lattice filter whose transfer function corresponding to FIR filter described by the transfer function

$$H(z) = 1 + 2z^{-1} + \frac{1}{3}z^{-2}. \text{ Also draw the corresponding lattice structure.} \quad 8$$

5. a) Derive the expression for poles from the squared magnitude response of Butterworth L.P.F. 10

- b) Determine the order of a Chebyshev filter to meet the following specifications :
 Passband ripple = – 2db at passband edge frequency 1 rad/sec and 20 db attenuation at stopband edge of 1.3 rad/sec. 10

6. a) Derive the transformation of IIR filter using approximation of derivatives by backward difference and verify whether it satisfies the sufficient and necessary conditions of mapping. 5

- b) Convert the following transfer function into digital using impulse invariance method. 5

$$H(s) = \frac{s+a}{(s+a)^2 + b^2}$$

- c) Design a digital filter $H(z)$ that when used as A/D – $H(z)$ – D/A structure gives an equivalent analog with the following specifications : 10

Passband ripple ≤ 3 db

Passband edge = 500 hz

Stopband attenuation ≥ 15 db

Stopband edge = 750 Hz

Sampling rate = 2 KHz

Use Bilinear transformation on butterworth prototype.



7. a) Derive the frequency response of a symmetric 7 IR low pass filter for both N even and N odd. 10
- b) Design an ideal low pass filter using Hamming window. The frequency response of the filter is $H_d(e^{j\omega}) = 1 - \pi/2 \leq \omega \leq \pi/2$.
 $= 0$ elsewhere
- Select the length of the unit impulse response of FIR filter as 9. 10
8. Write a brief note on the following :
- 1) Frequency sampling design of 7IR filters and its realisation in direct form structure. 5
 - 2) Linear convolution for filtering long sequences. 5
 - 3) TMS 320 DSP processors. 10
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