

**Fifth Semester B.E. Degree Examination, June/July 2013**  
**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. Consider two periodic sequences  $x(n)$  and  $y(n)$ .  $x(n)$  and  $y(n)$  has the period  $N$  and  $M$  respectively. The sequence  $w(n) = x(n) + y(n)$ . (i) Show that  $w(n)$  is periodic with period  $MN$ . (ii) Also show that  $w(k)$  represents  $MN$  point DFT of an  $MN$  point sequence  $w(n)$ . (06 Marks)
- b. Find the 4 point DFT of the sequence  $x(n) = \cos(n\pi/4)$ . (06 Marks)
- c. Obtain the relationship between (i) DFT and DTFT (ii) DFT and DFS. (08 Marks)
- 2 a. Let  $x(k)$  denote the  $N$  point DFT of the  $N$  point sequence  $x(n)$ .  
 (i) Show that if  $x(n)$  satisfies the relation  $x(n) = -x(N-1-n)$ , then  $x(0) = 0$   
 (ii) Show that when  $N$  even and if  $x(n) = x(N-1-n)$ , then  $x(N/2) = 0$ . (10 Marks)
- b. Compute the circular convolution using DFT and IDFT for the following sequence  
 $x_1(n) = \{2, 3, 1, 1\}$  and  $x_2(n) = \{1, 3, 5, 3\}$  (10 Marks)
- 3 a.  $g(n)$  and  $h(n)$  are two sequences of length 6. They have 6 point DFTS  $G(k)$  and  $H(k)$  respectively. Let  $g(n) = \{4.1, 3.5, 1.2, 5, 2, 3.3\}$ . The DTFS  $G(k)$  and  $H(k)$  are related by the circular frequency shift as  $H(k) = G((k-3))_6$ . Determine  $h(n)$  without computing DFT and IDFT. (08 Marks)
- b. Determine 8-point DFT of  $x(n) = \{1, 0, -1, 2, 1, 1, 0, 2\}$  using radix 2 DIT FFT algorithm. (12 Marks)
- 4 a. Compute DFT of two real sequences using FFT algorithms. (08 Marks)
- b. Explain Goertzel algorithm. (08 Marks)
- c. Discuss memory requirement and Inplace computation related to DIT and DIF FFTs. (04 Marks)

**PART – B**

- 5 a. Design a low pass 1 rad/sec bandwidth Chebyshev filter with the following characteristics:  
 (i) Acceptable pass band ripple of 2 dB  
 (ii) Cutoff radian frequency of 1 rad/sec  
 (iii) Stopband attenuation of 20 dB or greater beyond 1.3 rad/sec. (12 Marks)
- b. Convert the following low pass digital filter of cutoff frequency  $0.2\pi$  into high pass filter of cutoff frequency  $0.3\pi$  radians

$$H(z) = \frac{0.245(1 + z^{-1})}{1 - 0.509z^{-1}}$$

(08 Marks)

- 6 a. What are the properties of FIR filters? State their importance. (04 Marks)
- b. What is Gibbs phenomenon? How it can be reduced? (04 Marks)
- c. Determine the filter coefficients  $h_d(n)$  for the desired frequency response of a low pass filter

$$\text{given by, } H_d(e^{j\omega}) = \begin{cases} e^{-j2\omega} & \text{for } -\pi/4 \leq \omega \leq \pi/4 \\ 0 & \text{for } \pi/4 \leq |\omega| \leq \pi \end{cases}$$

If we define the new filter coefficients by  $h(n) = h_d(n) \cdot w(n)$

$$\text{where } w(n) = \begin{cases} 1 & \text{for } 0 \leq n \leq 4 \\ 0 & \text{elsewhere} \end{cases}$$

Determine  $h(n)$  and also frequency response  $H(e^{j\omega})$ .

(12 Marks)

- 7 a. Explain how an analog filter is mapped on to a digital filter using impulse invariance method. What are the limitations of the method? (08 Marks)
- b. Design a digital LPF 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  $|H(j\Omega)| > 60$  dB for  $\Omega \geq 8268\pi$  radian/sec. Sampling interval  $T_s = 10^{-4}$  sec use Bilinear transformation techniques for designing.

(12 Marks)

- 8 a. Develop the lattice ladder structure for the filter with difference equation

$$y(n) + \frac{3}{4}y(n-1) + \frac{1}{4}y(n-2) = x(n) + 2x(n-1)$$

(10Marks)

- b. For  $y(n) = -0.1y(n-1) + 0.2y(n-2) + 3x(n) + 3.6x(n-1) + 0.6x(n-2)$ , obtain direct form I and II and cascade form with single pole-zero subsystem. (10 Marks)