



**V Semester B.E. (Electrical and Electronics) Degree  
Examination, December 2017/January 2018  
(2K11 Scheme)  
EE 505 : DIGITAL SIGNAL PROCESSING**

Time : 3 Hours

Max. Marks : 100

**Instructions :** i) Normalized Filter Tables are **permitted** to use.  
ii) Answer **any five full** questions choosing at least **two** questions from Part **A** and Part **B**.

## PART – A

1. a) State and prove the following DFT properties : 10  
i) Time reversal of a sequence  
ii) Circular time shift of a sequence  
iii) Parseval's Theorem.
- b) Compute the 4-point circular convolution of the sequences given by  $x(n) = \{2, 1, 3, -2\}$  and  $h(n) = \{2, 9, 2, 9\}$  using DFT and IDFT method. 6
- c) Compute the N-point DFT of the sequence 4  
 $x(n) = \cos(n W_0)$ , where  $W_0 = \frac{2\pi}{N} K_0, 0 < n < N-1$ .
2. a) Find the DFT of the sequence  $x(n) = \{1, 2, -1, 2, 3, 4, 1, 2\}$  using Radix-2 DIT-FFT algorithm. 10  
b) Derive the Radix-2 DIF-FFT algorithm to compute DFT of an  $N = 8$  point sequence and draw the complete signal flow graph. 10
3. a) Write a short note on Goertzel Algorithm. 4  
b) Let  $x(n)$  be a finite length sequence with  $x(k) = \{0, 1 + j, 1, 1 - j\}$ . Find DFT's of the following sequences : 8  
i)  $x_1(n) = e^{j\pi/2^n} x(n)$                       iii)  $x_3(n) = x((n-1))_4$   
ii)  $x_2(n) = \cos(\pi/2^n) x(n)$                       iv)  $x_4(n) = (0, 0, 1, 0) \otimes_4 x(n)$ .

P.T.O.



- c) Let  $x(t)$  be an analog signal with bandwidth  $B = 3$  KHz.  $N = 2^m$  – point DFT is used to compute the spectrum of the signal with a resolution less than or equal to 50 Hz. Determine
- the minimum sampling rate
  - the minimum number of required samples
  - the minimum length of the analog signal record. **8**
4. a) Consider a second order LTI system described by the difference equation
- $$y(n) = \frac{1}{16} y(n-2) + x(n).$$
- Determine the direct form – I, direct form – II, parallel – form and cascade – form realizations of the system. **10**
- b) Realise an FIR filter with impulse response  $h(n)$  given by
- $$h(n) = \left(\frac{1}{2}\right)^n [u(n) - u(n-4)]$$
- using direct form – I. **4**
- c) Realize the linear-phase FIR filter having the following impulse response.
- $$h(n) = \delta(n) - \frac{1}{6} \delta(n-1) + \frac{1}{8} \delta(n-2) + \frac{1}{16} \delta(n-3) - \delta(n-4).$$
- 6**

## PART – B

5. a) Distinguish between Butterworth and Chebyshev Type I filter. **4**
- b) Design a Chebyshev I filter to meet the following specifications :
- Passband ripple :  $\leq 2$  dB
  - Passband edge : 1 rad/sec
  - Stopband attenuation :  $\geq 20$  dB
  - Stopband edge : 1.3 rad/sec
- Assume the use of lowpass Chebyshev I filter. **8**
- c) Design a Butterworth Analog highpass filter that will meet the following specifications :
- Maximum passband attenuation = 2 dB
  - Passband edge freq. = 200 rad/sec
  - Minimum Stopband attenuation = 20 dB
  - Stopband edge freq. = 100 rad/sec. **8**



6. a) Derive the necessary mathematical modeling for Bilinear transformation of designing IIR filters. Show the mapping of S-plane to z-plane. **10**
- b) A digital lowpass filter is required to meet the following specifications. **10**
- $20 \log | H(W) |_{w = 0.2 \pi} \geq - 1.9328 \text{ dB}$
- $20 \log | H(W) |_{w = 0.6 \pi} \leq - 13.9794 \text{ dB}$ . The filter must have a maximally flat frequency response. Find  $H(z)$  to meet above specifications using IIT (Impulse Invariance Transformation).
7. a) Design a digital filter  $H(z)$  that when used in an A/D –  $H(z)$  – D/A structure gives an equivalent filter with the specifications **10**
- Passband ripple :  $\leq 3.01 \text{ dB}$
- Passband edge : 500 Hz
- Stopband attenuation :  $\geq 20 \text{ dB}$
- Stopband edge : 750 Hz
- Sample rate : 4KHz
- Use bilinear transformation and Butterworth prototype.
- b) Explain the design of FIR digital filter using frequency sampling technique. Derive the mathematical equations and also draw the frequency sampling structure. **10**
8. a) Distinguish between IIR and FIR digital filters. **4**
- b) Obtain the coefficients of an FIR filter to meet the following specifications : **10**
- Passband edge freq. 1.5 KHz
- Stopband edge freq. : 2 KHz
- Minimum stopband Attenuation : 50 dB
- Sampling frequency : 8 KHz
- c) What are the advantages and disadvantages of window technique ? Also explain the design of FIR differentiators. **6**
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