Reg. No.:													

## Question Paper Code: 71735

## B.E./B.Tech. DEGREE EXAMINATION, APRIL/MAY 2017.

Fifth /Sixth Semester

Electronics and Communication Engineering

## EC 6502 — PRINCIPLES OF DIGITAL SIGNAL PROCESSING

(Common to Biomedical Engineering, Medical Electronics)

(Regulations 2013)

Time: Three hours

Maximum: 100 marks

Answer ALL questions.

PART A —  $(10 \times 2 = 20 \text{ marks})$ 

- 1. What is the relation between DTFT and DFT?
- 2. Compute the DFT of the sequence  $x(n) = \{1, -1, 1, -1\}$ .
- 3. What are the requirements for the digital filter to be stable and casual?
- 4. Discuss the need for prewarping.
- 5. What is Gibbs phenomenon?
- 6. Compare Hamming window with Blackmann window.
- 7. What are the methods used to prevent overflow?
- 8. What is meant by "dead band" of the filter?
- 9. Define adaptive filtering.
- 10. List the applications of multirate signal processing.

## PART B — $(5 \times 16 = 80 \text{ marks})$

11. (a) Compute the DFT for the sequence {1, 2, 3, 4, 4, 3, 2, 1}. Using radix - 2
DIF - FFT algorithm. (16)

Or

- (b) In an LTI system the input  $x(n) = \{1, 1, 2, 1\}$  and the impulse response  $h(n) = \{1, 2, 3, 4\}$ . Perform the circular convolution using DFT and IDFT. (16)
- 12. (a) Design a digital Butterworth filter with the following specifications  $0.707 \le \left|H(e^{jw})\right| \le 1,\ 0 \le \omega \le 0.5\pi$   $\left|H(e^{jw})\right| \le 0.2,\ 0.75\pi \le \omega \le \pi$

Determine system function H(z) for a Butterworth filter using Bilinear transformation. (16)

Or

- (b) Determine the system function of the lowest order digital Chebyshev filter with the following specifications, 3db ripple in the pass band  $0 \le \omega \le 0.2\pi$  and 25db attenuation in the stop band  $0.45 \pi \le \omega \le \pi$ . (16)
- 13. (a) Design a HPF with the following frequency response

$$H_d(e^{j\omega}) = 1$$
 for  $\pi/4 \le |\omega| \le \pi$   
= 0 for  $|\omega| \le \pi/4$ 

of length N = 11 using Hanning window.

(16)

Or

(b) Determine the coefficients of a linear phase FIR filter of length N = 15 which has a symmetric unit sample response and a frequency response that satisfies the conditions. (16)

$$H(2\pi k/15) = 1$$
; for  $k = 0, 1, 2, 3$   
= 0; for  $k = 4, 5, 6, 7$ 

14. (a) Two first order filters are connected in cascaded whose system functions of the individual sections are  $H1(z) = 1/(1 - 0.5z^{-1})$  and  $H2(z) = 1/(1 - 0.6z^{-1})$ . Determine the overall output noise power. (16)

Or

- (b) Explain the characteristics of limit cycle oscillations with respect to the system described by the difference equation y(n) = 0.95y(n-1) + x(n). Determine the dead band. (16)
- 15. (a) Explain the concept of deciation by a factor D and interpolation by factor I. With help of equation explain sampling rate conversion by a rational factor I/D. (16)

Or

(b) Explain the operation of adaptive filter with suitable diagrams and equations. (16)

71735