Reg. No.:				K.								10 -
-----------	--	--	--	----	--	--	--	--	--	--	--	------

Question Paper Code: 80343

B.E./B.Tech. DEGREE EXAMINATION, NOVEMBER/DECEMBER 2016.

Fifth Semester

Medical Electronics

EC 6502 — PRINCIPLES OF DIGITAL SIGNAL PROCESSING

(Common to Electronics and Communication Engineering and Sixth Semester Biomedical Engineering)

(Regulations 2013)

Time: Three hours

Maximum: 100 marks

Answer ALL questions.

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

- 1. Compare Radix 2 DIT, DIF FFT algorithm.
- 2. Test the causality and stability of $y(n) = \sin x(n)$.
- 3. What is known as prewarping?
- 4. What are the properties of bilinear transformation?
- 5. What do you understand by linear phase response?
- 6. What are the desirable characteristics of the window?
- 7. What are the different types of fixed point representation?
- 8. Name the three quantization error due to finite word length registers in digital filters.
- 9. What is the need for anti imaging filter after upsampling signal?
- 10. What is meant by adaptive filter?

PART B - (5 × 16 = 80 marks)

11. (a) Derive radix 2 – DIT FFT algorithm and obtain DFT of the sequence $x(n) = \{1, 2, 3, 4, 4, 3, 2, 1\}$ using DIT algorithm. (16)

Or

- (b) (i) Compute IDFT of the sequence $X(K) = \{7, -0.707, -j0.707, -j, 0.707 j0.707, 1, 0.707 + j0.707 j, -0.707 + j0.707\}$ using DIF algorithm. (10)
 - (ii) Perform the linear convolution of finite duration sequences $h(n) = \{1,2\}$ and $x(n) = \{1,2,-1,2,3,-2,-3,-1,1,2,-1\}$ by overlap save method. (6)
- 12. (a) Design a third order Butterworth digital filter using impulse invariant technique. Assume sampling period T = 1 sec. (16)

Or

- (b) Convert the single pole low pass filter with system function $H(z) = \frac{0.5\left(1+z^{-1}\right)}{1-0.302z^{-2}} \text{ into band pass filter with upper and lower cut off}$ frequencies w_u & w_L respectively. The LPF has 3dB BW of $w_p = \frac{\pi}{6}$ & $w_u = \frac{3\pi}{4}$, $w_l = \frac{\pi}{4}$.
- 13. (a) Design an ideal BPF with a frequency response $H\alpha(e^{jw})=1, \quad for \frac{\pi}{4} \leq |w| \leq \frac{3\pi}{4}.$ 0, otherwise

Find the value of h(n) for N=11 and plot the frequency response. (16)

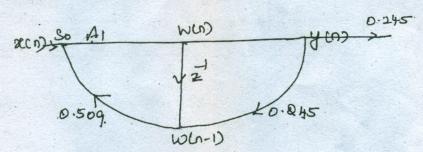
Or

- (b) Design a linear phase FIR filter with a cut off frequency of $\frac{\pi}{2} r/\sec$.

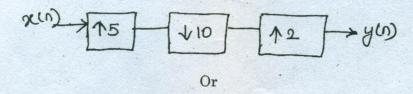
 Take N = 17 using frequency sampling techniques. (16)
- 14. (a) Study the limit cycle behaviour of the system described by y(n) = Q[a y(n-1)] + x(n), where y(n) is the output of the filter and Q[.] is quantization. Assume $a = \frac{7}{8}$, $x(0) = \frac{3}{4}$ & x = 0, for n > 0 choose 4 bit sign magnitude.

Or

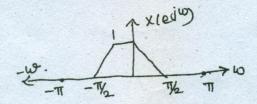
(b) For the digital network shown in figure find H(z) and scale factor. So to avoid over flow register A_1 (16)



- 15. (a) (i) Explain in about detail the multistage implementation of sampling rate conversion. (8)
 - (ii) For the multirate system shown in figure develop an expression for the output y(n) as a function of i/p x(n) (8)



- (b) (i) Show that the upsampler and down sampler are time variant systems. (8)
 - (ii) The frequency response of x(n) is shown in figure



If the input is passed through a down sampler by 2, find the frequency response of output and give your comment on aliasing. (8)