



III Semester M.Sc. Degree Examination, December 2014
ELECTRONIC SCIENCE
EL – 302 : Digital Signal Processing

Time : 3 Hours

Max. Marks : 80

Instructions : 1) *Section-A* : Answer **all** questions from *Section A*.
2) *Section-B* : Answer **any four full** questions from *Section B*.

SECTION – A

(5×4=20)

1. Find the Z-transform of positive time exponential sequence.
2. Compute the 4 point DFT of sequence $x(n) = 1^n$ $0 \leq n \leq 3$.
3. Explain the concept of Power density spectrum.
4. Draw and explain how the order of the filter affects the frequency response of a low pass filter using Butterworth approximation.
5. Explain the concept of Joint time frequency analysis.

SECTION – B

6. a) Determine the step response of a causal LTI system using Z-Transform
 $y(n) = \alpha y(n-1) + x(n)$; $-1 < \alpha < 1$. 6
b) Determine the Z-Transform of the sequence $x(n) = \sin(\omega n)$. 5
c) Find the inverse Z-Transform of $X(Z) = Z^2(1 - \frac{1}{2}Z^{-1})(1 + Z^{-1})(1 - Z^{-1})$. 4
7. a) Given $x[n] = [4 \quad \underset{\uparrow}{3} \quad 2 \quad -1 \quad 5 \quad 2 \quad 3]$.
Determine :
i) $y_1[n] = x[2n + 3]$ ii) $y_2(n) = x(-n - 3)$. 4
b) Given the 2 systems $h_1[n] = [2 \quad 1 \quad 0.5]$ and $h_2[n] = [2 \quad 2 \quad 1 \quad 1]$
are connected in parallel. Determine the output of the system if its input is
 $x[n] = [1 \quad 2 \quad -2]$. 5
c) Write a note on classification of signals with examples. 6

P.T.O.



8. a) Find DFT of the sequence $x(n) = n + 1 ; 0 \leq n \leq 7$, using FFT algorithm. **10**
 b) Mention the difference between FIR and IIR filter. **5**
9. a) It is required to meet the following specifications of a LPF using Butterworth approximation. Determine the filter coefficients. **10**
 Pass band : $|H(j\omega)| \geq -2\text{dB} ; 0 \leq \omega \leq 0.3\pi$
 Stop band : $|H(j\omega)| \leq -60\text{dB} ; \omega \geq 0.45\pi$.
- b) Determine the digital transfer function using bilinear transformation for the system $H(s) = \frac{2}{s^2 + 3s + 2}$ at (i) $T = 1$ sec and (ii) $T = 0.1$ sec. **5**
10. a) Design a high pass filter using the frequency sampling method. The cut off freq. is $\frac{\pi}{4}$. Apply a 8 point rectangular window. **10**
- b) Explain the quantization error. Derive the analytical expression for SQNR. **5**
11. Write short notes on the following : **(3×5=15)**
 a) Properties of DSP Systems
 b) In-place computation
 c) Error sources in spectrum analysis using DFT
 d) Comparison of analog and digital signal processing systems
 e) State and prove the multiplication property of DFT.
-