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**ADVANCED DIGITAL SIGNAL PROCESSING  
(ELECTIVE – I)****Course Code:13EE2206****L P C  
4 0 3****Pre requisites:** Signals and Systems and Digital Signal Processing.**Course Outcomes:** At the end of the course, the student will be able to

CO1: Demonstrate realization of different structures of IIR and FIR Filters.

CO2: Describe Spectral transformations of IIR filters

CO3: Design computationally efficient FIR digital filters

CO4: Identify Quantization process, arithmetic round off and signal-to-noise ratio in low order IIR filters

CO5: Analyze errors and Estimate power spectrum using various methods

**UNIT-I****DIGITAL FILTER STRUCTURES:**

Block diagram representation-Equivalent Structures- IIR digital filter Structures All Pass Filters-tunable IIR Digital Filters-IIR tapped cascaded Lattice Structures--Computational complexity of digital filter structures.

Block diagram representation- Equivalent structure- FIR digital Filter Structures FIR cascaded Lattice Structures – Parallel- Digital Sine cosine generator-Computational complexity of digital filter structures.

**UNIT-II****IIR DIGITAL FILTER DESIGN**

Preliminary considerations-Bilinear transformation method of IIR filter design-design of Low pass, high pass, Band pass, and Band stop- IIR digital filters-Spectral transformations of IIR filters.

**UNIT-III****FIR DIGITAL FILTER DESIGN**

FIR filter design-based on Windowed Fourier series- computer aided design of Equiripple Linear Phase FIR filters- design of minimum phase FIR filters- design of computationally efficient FIR digital filters

**UNIT-IV****DSP ALGORITHM AND FINITE WORD LENGTH EFFECTS**

Computation of the discrete Fourier transform- Number representation- Arithmetic Operations-handling of overflow-Tunable digital filters-function approximation.

The Quantization process and errors- Quantization of fixed -point and floating -point Numbers-Analysis of coefficient Quantization effects - Analysis of Arithmetic Round-off errors-Dynamic range scaling signal-to-noise ratio in Low -order IIR filters-Low-Sensitivity Digital filters

**UNIT- V****ERRORS ANALYSIS AND POWER SPECTRUM ESTIMATION**

Reduction of Product round-off errors using error feedback-Limit cycles in IIR digital filters- Round-off errors in FFT Algorithms.

Estimation of spectra from Finite Duration Observation of signals – Non-parametric methods for power spectrum Estimation–parametric method for power spectrum Estimation-Estimation of spectral form-Finite duration observation of signals-Non-parametric methods for power spectrum estimation-Welch methods-Blackman & Tukey method.

**TEXT BOOKS:**

1. Sanjit K. Mitra , “*Digital signal processing*”,2nd Edn, TMH, 1997 (UNITS – I, II, III, IV)
2. John G.Proakis, “*Digital Signal Processing principles, algorithms and Applications*”, 3rd Edn. PHI, 2002. (UNITS – V)

**REFERENCE BOOKS:**

1. Alan V.Oppenheim, Ronald W.Shafer, “*Discrete Time Signal Processing*”, 1<sup>st</sup> Edition. PHI, 1996.
2. S.Salivahanan, A.Vallavaraj, C. Gnanapriya , “*Digital Signal Processing*“, 2<sup>nd</sup> Edition, TMH, 2001
3. Lourens R. Rabinar & Bernard Gold, “*Theory and Applications of Digital Signal Processing*“, 2<sup>nd</sup> Edition, TMH, 2001
4. Andreas Antoniou, “*Digital Filter Analysis and Design*”, TMH, 2001