ADVANCED DIGITAL SIGNAL PROCESSING (ELECTIVE – I)

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 (10-Lectures)

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 (10-Lectures)

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 (10-Lectures)

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 (10-Lectures)

DSP ALGORIRHM 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 (10-Lectures)

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", 1stEdition. PHI, 1996.
- 2. S. Salivahanan, A. Vallavaraj, C. Gnanapriya, "Digital Signal Processing", 2ndEdition, TMH, 2001
- 3. Lourens R. Rabinar& Bernard Gold, "Theory and Applications of Digital Signal Processing", 2ndEdition, TMH, 2001
- 4. Andreas Antoniou, "Digital Filter Analysis and Design", TMH, 2001