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By K B Hemanth Raj

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ADAPTIVE SIGNAL PROCESSING

B.E., VI Semester, Electronics & Communication Engineering/ Telecommunication Engineering [As per Choice Based Credit System (CBCS) Scheme]

Course Code	17EC652	CIE Marks	40
Number of Lecture	03	SEE Marks	60
Hours/Week			
Total Number of	40 (8 Hours / Module)	Exam	03
Lecture Hours		Hours	

CREDITS - 03

Course Objectives: The objectives of this course are to:

- Introduce to the concept and need of adaptive filters and popular adaptive signal processing algorithms
- Understand the concepts of training and convergence and the trade-off between performance and complexity.
- Introduce to common linear estimation techniques
- Demonstrate applications of adaptive systems to sample problems.
- Introduce inverse adaptive modelling.

Module-1

Adaptive systems: Definitions and characteristics - applications – properties-examples - adaptive linear combiner input signal and weight vectors - performance function-gradient and minimum mean square error - introduction to filtering-smoothing and prediction - linear optimum filtering-orthogonality - Wiener – Hopf equation-performance surface(Chapters 1& 2 of Text). **L1, L2**

Module-2

Searching performance surface-stability and rate of convergence: Learning curvegradient search - Newton's method - method of steepest descent - comparison - Gradient estimation - performance penalty - variance - excess MSE and time constants - mis-adjustments (Chapters 4& 5 of Text). **L1, L2**

Module-3

LMS algorithm convergence of weight vector: LMS/Newton algorithm - properties - sequential regression algorithm - adaptive recursive filters - random-search algorithms - lattice structure - adaptive filters with orthogonal signals (Chapters 6 & 8 of Text).

L1, L2, L3

Module-4

Applications-adaptive modeling and system identification: Multipath communication channel, geophysical exploration, FIR digital filter synthesis. (Chapter 9 of Text). **L1**, **L2**, **L3**

Module-5

Inverse adaptive modeling: Equalization, and deconvolution adaptive equalization of telephone channels-adapting poles and zeros for IIR digital filter synthesis (Chapter 10 of Text). **L1**, **L2**, **L3**

Course Outcomes: At the end of the course, students should be able to:

- Devise filtering solutions for optimising the cost function indicating error in estimation of parameters and appreciate the need for adaptation in design.
- Evaluate the performance of various methods for designing adaptive filters through estimation of different parameters of stationary random process clearly considering practical application specifications.

- Analyse convergence and stability issues associated with adaptive filter design and come up with optimum solutions for real life applications taking care of requirements in terms of complexity and accuracy.
- Design and implement filtering solutions for applications such as channel equalisation, interference cancelling and prediction considering present day challenges.

Text Book:

Bernard Widrow and Samuel D. Stearns, "Adaptive Signal Processing", Person Education, 1985.

Reference Books:

- 1. Simon Haykin, "Adaptive Filter Theory", Pearson Education, 2003.
- 2. John R. Treichler, C. Richard Johnson, Michael G. Larimore, "Theory and Design of Adaptive Filters", Prentice-Hall of India, 2002.