## SEMESTER 5

# ELECTRONICS & COMMUNICATION ENGINEERING

## **ELECTROMAGNETICS**

Course Code	PCECT501	CIE Marks	40
Teaching Hours/Week (L: T:P: R)	3:1:0:0	ESE Marks	60
Credits	4	Exam Hours	2Hrs. 30 Min.
Prerequisites (if any)	Physics for Electrical sciences (GBPHT121)	Course Type	Theory

## **Course Objectives:**

1. To impart knowledge on the basic concepts of electric and magnetic fields and its applications.

Module No.	Syllabus Description	Contact Hours
1	Review of coordinate system-Rectangular, cylindrical and spherical coordinate systems. Review of vector calculus- curl, divergence gradient. Review of Coulomb's law, Gauss's law and Ampere's current law. Derivation of capacitance and inductance of two wire transmission line and coaxial cable. Magnetic scalar and vector potential. Poisson and Laplace equations, Determination of voltage and electric field using Laplace and Poisson's equation.	12
2	Maxwell's equation from fundamental laws. Boundary condition of electric field and magnetic field from Maxwell's equations. Solution to wave equation Propagation of plane EM wave in perfect dielectric, lossy medium, good conductor, skin depth. Polarization of waves.	10
3	Reflection and refraction of plane electromagnetic waves at boundaries for normal & oblique incidence (parallel and perpendicular polarization), Snell's law of refraction, Brewster angle. Power density of EM wave, Poynting	10

	vector theorem.	
4	Transmission line as circuit elements (L and C). Transmission line equations and characteristic impedance. Reflection coefficient and VSWR. Derivation of input impedance of transmission line. Calculation of line impedance and VSWR using smith chart. The hollow rectangular waveguide –TE and TM wave-dominant mode, group velocity and phase velocity –derivation and simple problems only.	12

(CIE: 40 marks, ESE: 60 marks)

## **Continuous Internal Evaluation Marks (CIE):**

Attendance	Assignment/ Microproject	Internal Examination-1 (Written)	Internal Examination- 2 (Written)	Total
5	15	10	10	40

## **End Semester Examination Marks (ESE)**

Part A	Part B	Total
2 Questions from each	Each question carries 9 marks.	
module.	Two questions will be given from each module, out of	
• Total of 8 Questions, each	which 1 question should be answered.	
carrying 3 marks	Each question can have a maximum of 3 sub	60
	divisions.	
(8x3 =24 marks)	(4x9 = 36  marks)	

At the end of the course students should be able to:

	Course Outcome	Bloom's Knowledge Level (KL)
CO1	Summarize the basic mathematical concepts related to electromagnetic vector fields.	K2
CO2	Apply Maxwell's equations in different forms to diverse electromagnetic problems.	К3
CO3	Analyze reflection, refraction and power density of electromagnetic waves.	К3
CO4	Analyse the propagation of EM waves in transmission lines and wave guides.	К3

Note: K1- Remember, K2- Understand, K3- Apply, K4- Analyse, K5- Evaluate, K6- Create

	PO1	PO2	PO3	PO4	PO5	PO6	PO7	PO8	PO9	PO1 0	PO1 1	PO1
CO1	3											2
CO2	3	3	2	2	2							2
CO3	3	3	2	2	2							2
CO4	3	3	2	2	2							2

	Text Books							
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year				
1	Elements of Electromagnetics	Matthew N. O. Sadiku	Oxford University Press	7 <sup>th</sup> edition, 2018				
2	Engineering Electromagnetics	William Hayt and John Buck	McGraw-Hill Higher Education	9 <sup>th</sup> edition, 2019				
3	Electromagnetic Waves and Transmission Lines	Y Mallikarjuna Reddy	The Orient Blackswan	1 <sup>st</sup> edition 2015				

	Reference Books							
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year				
1	Schaum's Outline of Eletromagnetics	Mahmood Nahvi; Joseph Edminister	McGraw-Hill	5 <sup>th</sup> edition, 2019				
2	Engineering Electromagnetics Essentials	B N Basu	The Orient Blackswan	1 <sup>st</sup> edition 2015				

Video Links (NPTEL, SWAYAM)				
Module No.	Link ID			
1	https://onlinecourses.nptel.ac.in/noc21_ee83/preview			
2	https://onlinecourses.nptel.ac.in/noc21_ee83/preview			
3	https://nptel.ac.in/courses/115101005			
4	https://archive.nptel.ac.in/courses/117/101/117101056/#			

## ANALOG AND DIGITAL COMMUNICATION

Course Code	PCECT502	CIE Marks	40
Teaching Hours/Week	3-1-0-0	ESE Marks	60
(L: T:P: R) Credits	4	Exam Hours	2Hrs. 30 Min.
Prerequisites (if any)	PCECT402 Signals and Systems GBMAT401 Probability, Random Process and Numerical Methods	Course Type	Theory

## **Course Objectives:**

1. To analyse different analog and digital communication systems

Module	Syllabus Description	Contact Hours
No.		
	Block diagram of a communication system. Need for modulation. Amplitude	
	modulation, Equation and spectrum of AM signal, DSB-SC, SSB -pilot	
	carrier and Vestigial sideband systems.	
1	Angle modulation: Narrow and wide band FM and their spectra, relationship	
1	between FM and PM, Carson's rule, pre-emphasis and de-emphasis filtering.	12
	Comparison of AM and FM, Block diagram of FM receiver.	12
	Superheterodyne receivers- Characteristics of receivers –image frequency.	
	Noise: external, internal, White noise.	
	Sampling and Quantization, SQNR for uniform quantization, Companding	
2	Pulse code modulation, Transmitter and receiver. DPCM transmitter and	10
	receiver. Delta modulation, Slope overload, Line codes.	
	Baseband data transmission of digital data through AWGN channel,	
	Mathematical model of ISI, Nyquist criterion for zero ISI, Signal modelling	
2	for ISI, Raised cosine spectrum, Equalization, Zero forcing Equaliser.	12
3	Geometric representation of Signals-Gram-Schmitt procedure, Signal space.	12
	Vector model of AWGN channel.	
	Matched filter and correlation receivers, MAP receiver, Maximum likelihood	

	receiver.	
4	Digital band pass modulation schemes-BPSK system and signal constellation. BPSK transmitter and receiver. QPSK system and Signal constellations. BER analysis of BPSK and QPSK in erfc. Plots of BER Vs SNR. QPSK transmitter and receiver. Quadrature amplitude modulation and signal constellation.	10
	signal constellation.	

(CIE: 40 marks, ESE: 60 marks)

## **Continuous Internal Evaluation Marks (CIE):**

Attendance	Assignment/ Microproject	Internal Examination-1 (Written)	Internal Examination- 2 (Written)	Total
5	15	10	10	40

## **End Semester Examination Marks (ESE)**

Part A	Part B	Total
2 Questions from each	Each question carries 9 marks.	
module.	Two questions will be given from each module, out	
• Total of 8 Questions, each	of which 1 question should be answered.	<b>60</b>
carrying 3 marks	• Each question can have a maximum of 3 sub	60
	divisions.	
(8x3 =24 marks)	(4x9 = 36  marks)	

At the end of the course students should be able to:

	Course Outcome	Bloom's Knowledge Level (KL)
CO1	Illustrate the principles of analog communication systems	K2
CO2	Explain the basic concepts of digital communication	K2
СОЗ	Analyse the baseband transmission of digital data through AWGN channel	К3
CO4	Apply various digital modulation techniques in the design of digital communication systems	К3

Note: K1- Remember, K2- Understand, K3- Apply, K4- Analyse, K5- Evaluate, K6- Create

	PO1	PO2	PO3	PO4	PO5	PO6	PO7	PO8	PO9	PO10	PO11	PO12
CO1	3	-	-	-	2	-	-	-	-	-	-	-
CO2	3	-	-	-	2	-	-	-	-	-	-	-
CO3	3	3	3	3	2	-	-	-	-	-	-	2
CO4	3	3	3	3	2	-	-	-	-	-	-	2

	Text Books						
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year			
1	Communication Systems	Simon Haykin and Michael Moher	Wiley	5th Edition, 2020			
2	Modern Digital and Analog Communication Systems	B.P. Lathi and Zhi Ding	Oxford University Press	5th Edition, 2018			
3	Introduction to Analog and Digital Communication, An Indian adaptation	Simon Haykin and Michael Moher	Wiley	2nd Edition, 2022			

Reference Books							
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year			
1	Principles of Communication Systems	Herbert Taub and Donald L. Schilling	McGraw-Hill Education	4th Edition, 2013			
2	Digital Communications	John G. Proakis and Masoud Salehi	McGraw-Hill Education	6th Edition, 2020			
3	Communication Systems Engineering	John G. Proakis and Masoud Salehi	Pearson	2nd Edition, 2001			
4	Digital Communications Systems, An Indian Adaptation	Simon Haykin	John Wiley& Sons	4 <sup>th</sup> Edition, 2021			
5	Electronic communication systems	George Kennedy	McGraw Hill	6th Edition, 2017			
6	Introduction to Digital Communications	Wayne Stark	Cambridge University Press	1st edition 2023			

	Video Links (NPTEL, SWAYAM)					
Module No.	Link ID					
1	https://youtu.be/hTAlcrqjNps?si=okoRHdUegx9pbOz3					
2	https://youtu.be/s_vmLqT_6NQ?si=MF2OW6AaICiYKTfj					

#### **CONTROL SYSTEMS**

Course Code	PCECT503	CIE Marks	40
Teaching Hours/Week (L: T:P: R)	3-0-0-0	ESE Marks	60
Credits	3	Exam Hours	2Hr. 30 Min.
Prerequisites (if any)	GBMAT301  Mathematics for Electrical Science -3	Course Type	Theory

## **Course Objectives:**

- 1. To study the elements of control system, modelling and perform stability analysis of systems.
- 2. To design control systems with compensating techniques.
- **3.** To understand the state variable analysis method.

Module	Syllabus Description	Contact		
No.	Synabus Description			
	Introduction: Basic Components of Control Systems, Open-Loop and			
1	Closed-Loop Control Systems with examples.			
	Mathematical modelling of control systems: Electrical Systems and			
	Mechanical translational systems.	8		
	Transfer Function: Block diagram reduction techniques, Signal flow	o l		
	graph, Mason's gain formula.			
	Time Domain Analysis of Control Systems: Standard Test signals, Time			
	response of first order systems (unit impulse, step and ramp inputs) and			
2	second order systems (step input only). Time response of undamped, under			
2	damped, critically damped second order system to unit step signal, Time	8		
	domain specifications for a second order underdamped system, Steady state	o l		
	error and static error coefficients.			
	Stability of linear control systems: Concept of BIBO stability, absolute			
3	stability, Routh Hurwitz Criterion.			
	Root Locus Techniques: Introduction, properties and its construction.			

		Frequency domain analysis: Frequency domain specifications					
		Relative stability: gain margin and phase margin. Stability analysis using					
		Bode plot and Nyquist stability criterion.					
		P, PI & PID controllers: Introduction.					
		Design of Compensators: Need for compensators, lag and lead					
		compensators using Bode plots(only design steps)					
İ		State Variable Analysis of Linear Systems:					
		State variables, state equations, state variable representation of electrical					
	4	systems. Transfer function from State equation, Solutions of the state					
		equations, state transition matrix, Controllability and observability -	8				
		Kalman's Test.					

(CIE: 40 marks, ESE: 60 marks)

## **Continuous Internal Evaluation Marks (CIE):**

Attendance	Assignment/ Microproject	Internal Examination-1 (Written)	Internal Examination- 2 (Written)	Total	
5	15	10	10	40	

## **End Semester Examination Marks (ESE)**

Part A	Part B	Total
• 2 Questions from each	Each question carries 9 marks.	
module.	Two questions will be given from each module, out	
• Total of 8 Questions, each	of which 1 question should be answered.	(0
carrying 3 marks	• Each question can have a maximum of 3 sub	60
	divisions.	
(8x3 =24marks)	(4x9 = 36  marks)	

At the end of the course students should be able to:

	Course Outcome	Bloom's Knowledge Level (KL)
CO1	Analyze the systems using transfer function approach	К3
CO2	Perform time domain analysis and steady state analysis of systems	К2
CO3	Determine the absolute stability and relative stability of a system using Routh Hurwitz Criterion and root locus	К3
CO4	Apply frequency domain techniques to assess the system stability and to design different compensation techniques	К3
CO5	Analyse system Controllability and Observability using state space representation	К3

Note: K1- Remember, K2- Understand, K3- Apply, K4- Analyse, K5- Evaluate, K6- Create

	PO1	PO2	PO3	PO4	PO5	PO6	PO7	PO8	PO9	PO10	PO11	PO12
CO1	3	3	2		2							2
CO2	3	3	2		2							2
CO3	3	3	3		2							2
CO4	3	3	3		2							2
CO5	3	3	3		2							2

	Text Books					
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year		
1	Control Systems Engineering	I.J. Nagarath, M. Gopal	New Age International Publishers	7th Edition 2022		
2	Automatic Control Systems	Benjamin C. Kuo, Farid Golnaraghi,	Wiley	10th Edition 2017		
3	Modern Control Engineering	Katsuhiko Ogata	Pearson	Fifth Edition 2015		

	Reference Books					
Sl. No	Title of the Book  Name of the Author/s  Publisher		Edition and Year			
1	Feedback and Control Systems	Joseph DiStefano, Allen R. Stubberud, and Ivan J. Williams	McGraw Hill	Third Edition 2013		
2	Control systems	Ashok Kumar	Tata McGraw-Hill	Second Edition 2010		
3	Control Systems: Principles and Design	M Gopal	McGraw Hill Education	Fourth Edition 2012		
4	Nise's Control Systems Engineering	Norman S. Nise	Wiley India	8th Edition 2017		

	Video Links (NPTEL, SWAYAM)				
Module No.	Link ID				
1	https://youtu.be/Cl23xQrvFhk?feature=shared https://youtu.be/fsxSst10_cE?feature=shared				
2	https://youtu.be/cLyT6OWcmyU?feature=shared				
3	https://youtu.be/CZL7_Z0i1KQ?feature=shared				
4	https://youtu.be/CrXOMBlYFp0?feature=shared				

## **DIGITAL SIGNAL PROCESSING**

Course Code	PBECT504	CIE Marks	60
Teaching Hours/Week (L: T:P: R)	3:0:0:1	ESE Marks	40
Credits	4	Exam Hours	2 Hrs. 30 Min.
Prerequisites (if any)	Signals & Systems PCECT402	Course Type	Theory

## **Course Objectives:**

- 1. To describe signals mathematically and understand how to perform mathematical operations on signals
- 2. To gain knowledge of Digital filters

Module	Syllabus Description	
No.		
	Review of sampling, Z-Transform and DTFT	
	The Discrete Fourier Transform: DFT as a linear transformation (Matrix	
1	Relation), IDFT, Properties of DFT and examples (proof not necessary).	
1	Circular convolution, linear convolution using circular convolution, Filtering	9
	of long data sequences, overlap save and overlap add methods.	
	Frequency Analysis of Signals using the DFT (concept only required)	
	Design of FIR Filters - Symmetric and Anti-symmetric FIR Filters, Design	
	of linear phase FIR filters using Window methods, (rectangular, Hamming	
2	and Hanning). Design of IIR Digital Filters from Analog Filters	
	(Butterworth), IIR Filter Design by Impulse Invariance, and Bilinear	9
	Transformation, Frequency Transformations in the Analog Domain.	
	Structures for the realization of Discrete-Time Systems - Block diagram and	
	signal flow graph representations of filters.	
2	FIR Filter Structures: Linear structures, Direct Form.	9
3	IIR Filter Structures: Direct Form, Transposed Form, Cascade Form and	9
	Parallel Form.	
	Multi-rate Digital Signal Processing:	

	Decimation and Interpolation (Time domain and Frequency Domain	
	Interpretation ), Anti- aliasing and anti-imaging filter.	
	Efficient Computation of DFT: Fast Fourier Transform and computational	
	advantage over DFT, Radix-2 Decimation in Time FFT Algorithm.	
	Computer architecture for signal processing: Harvard Architecture,	
4	pipelining, MAC, Introduction to TMS320C67xx digital signal processor,	0
	Functional Block Diagram.	9
	Finite word length effects in DSP systems: Introduction, fixed-point and	
	floating-point DSP arithmetic, ADC quantization noise.	

#### **Suggestion on Project Topics**

Projects can include but not limited to, analysing various signals/finding their transforms and designing filters for extracting different frequency components. Projects can be simulated or implemented in hardware.

#### **Course Assessment Method**

(CIE: 60 marks, ESE: 40 marks)

#### **Continuous Internal Evaluation Marks (CIE):**

Attendance	Project	Internal Ex-1	Internal Ex-2	Total
5	30	12.5	12.5	60

#### **End Semester Examination Marks (ESE)**

Part A	Part B	Total
2 Questions from each	• 2 questions will be given from each module,	
module.	out of which 1 question should be answered.	
• Total of 8 Questions,	Each question can have a maximum of 2 sub	40
each carrying 2 marks	divisions. Each question carries 6 marks.	
(8x2 =16 marks)	(4x6 = 24  marks)	

At the end of the course students should be able to:

	Course Outcome		
CO1	Illustrate fundamental properties and relations relevant to DFT and solve basic problems involving DFT-based filtering methods.	K2	
CO2	Design linear phase FIR filters and IIR filters of different specifications.	К3	
CO3	Realise the various FIR and IIR filter structures for a given system function.	К3	
CO4	Compute DFT efficiently using FFT method and to explain the architecture of a DSP processor.	K2	

Note: K1- Remember, K2- Understand, K3- Apply, K4- Analyse, K5- Evaluate, K6- Create

	PO1	PO2	PO3	PO4	PO5	PO6	PO7	PO8	PO9	PO10	PO11	PO12
CO1	3	3	2		2							2
CO2	3	3	3		3							2
CO3	3	3	3		3							2
CO4	3	3	2		3							2

	Text Books					
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year		
1	Digital Signal Processing using Matlab	Vinay K. Ingle, John G. Proakis	Cengage Learning	3 <sup>rd</sup> Ed., 2011		
2	Think DSP: Digital Signal Processing using Python	Allen B. Downey	Green Tea Press	2 <sup>nd</sup> Ed., 2012		
3	Discrete-Time Signal Processing	Alan V Oppenheim, Ronald W. Schafer	Pearson Education	3 <sup>rd</sup> Ed., 2014		

	Reference Books					
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year		
1	Digital Signal Processing	Shaila D. Apte	Wiley	2nd Ed, 2019		
2	Digital Signal Processing: A Computer based Approach	Mitra S. K.	McGraw Hill	4 <sup>th</sup> Ed., 2014		
3	Digital Signal Processing: A Practical Approach	Ifeachor E. C., Jervis B. W.	Pearson Education	2 <sup>nd</sup> Ed., 2009		
4	Digital Signal Processing	Salivahanan S.	McGraw Hill	4 <sup>th</sup> Ed., 2019		

	Video Links (NPTEL, SWAYAM)					
Module No.	Link ID					
1	https://nptel.ac.in/courses/117102060 https://nptel.ac.in/courses/108105055					
2	same as above					
3	same as above					
4	same as above					

**PBL Course Elements** 

L: Lecture	R: Project (1 Hr.), 2 Faculty Members				
(3 Hrs.)	Tutorial	Practical	Presentation		
Lecture delivery	Project identification	Simulation/ Laboratory Work/ Workshops	Presentation (Progress and Final Presentations)		
Group discussion	Project Analysis	Data Collection	Evaluation		
Question answer Sessions/ Brainstorming Sessions	Analytical thinking and self-learning	Testing	Project Milestone Reviews, Feedback, Project reformation (If required)		
Guest Speakers (Industry Experts)	Case Study/ Field Survey Report	Prototyping	Poster Presentation/ Video Presentation: Students present their results in a 2 to 5 minutes video		

## Assessment and Evaluation for Project Activity

Sl. No	Evaluation for	
1	Project Planning and Proposal	5
2	Contribution in Progress Presentations and Question Answer Sessions	4
3	Involvement in the project work and Team Work	3
4	Execution and Implementation	10
5	Final Presentations	5
6	Project Quality, Innovation and Creativity	3
	Total	30

#### 1. Project Planning and Proposal (5 Marks)

- Clarity and feasibility of the project plan
- Research and background understanding
- Defined objectives and methodology

#### 2. Contribution in Progress Presentation and Question Answer Sessions (4 Marks)

- Individual contribution to the presentation
- Effectiveness in answering questions and handling feedback

#### 3. Involvement in the Project Work and Team Work (3 Marks)

- Active participation and individual contribution
- Teamwork and collaboration

#### 4. Execution and Implementation (10 Marks)

- Adherence to the project timeline and milestones
- Application of theoretical knowledge and problem-solving
- Final Result

#### 5. Final Presentation (5 Marks)

- Quality and clarity of the overall presentation
- Individual contribution to the presentation
- Effectiveness in answering questions

#### 6. Project Quality, Innovation, and Creativity (3 Marks)

- Overall quality and technical excellence of the project
- Innovation and originality in the project
- Creativity in solutions and approaches

#### **BIOMEDICAL ENGINEERING**

Course Code	PEECT521	CIE Marks	40
Teaching Hours/Week (L: T:P: R)	3-0-0-0	ESE Marks	60
Credits	3	Exam Hours	2 Hrs. 30 min.
Prerequisites (if any)	None	Course Type	Theory

## **Course Objectives:**

- 1. This course will introduce the various aspects of biomedical engineering and its applications described using engineering principles
- **2.** The student will be able to understand the techniques and uses of modern diagnostic and therapeutic equipment.

Module	Syllabus Description	Contact
No.	Synabus Description	
	Introduction to bio-medical engineering,	
	Sources of bio-electric potential: Resting and action potential, propagation of	
	action potentials. Various bioelectric potentials (ECG, EEG, EMG, ERG,	
	EOG, EGG concept only.)	
1	Electrode theory: Nernst equation, Electrode skin interface	
	Bio-potential electrodes: Microelectrodes, skin surface electrodes, needle	9
	electrodes	
	Bio-potential amplifiers: instrumentation amplifiers, carrier amplifiers,	
	isolation amplifiers, chopper amplifiers	
	Heart and cardiovascular system: electro conduction system of the heart,	
	ECG lead configurations, Einthoven triangle, Electrocardiography, ECG	
2	machine - block diagram, ECG recording system.	
	The human nervous system: Neurons, action potential of brain, brain waves,	9
	placement of electrodes, EEG recording, evoked potential,	

	Electrical activity of muscles: EMG signal acquisition and analysis.	
	Myoelectric control system. Electrical stimulation of the muscle and nerve,	
	Applications of EMG	
	Instruments for clinical laboratory: Oxymeters, blood cell counter, flame	
	photometer, Spectrophotometer	
	Therapeutic Equipments: Principles, block schematic diagram, working and	
3	applications of pacemakers, cardiac defibrillators, heart-lung machine,	9
3	dialyzers, surgical diathermy equipment, ventilators	9
	Biomedical Telemetry system: Components of biotelemetry system,	
	application of telemetry in medicine, single channel telemetry system for	
	ECG.	
	Medical Imaging systems (Basic Principle only): X-ray imaging - X-ray	
	machine, applications of X-rays in medicine.	
	Computed Tomograpy: Principle, image reconstruction, scanning system and	
	applications	
4	Ultrasonic imaging systems: Basic pulse echo system, Different types of	
	Ultrasonics systems:, A-Scan, B-Scan, M-Scan, applications, real-time	9
	ultrasonic imaging systems and probes.	
	Magnetic Resonance Imaging – Basic NMR components, Biological effects	
	and advantages of NMR imaging	

(CIE: 40 marks, ESE: 60 marks)

## **Continuous Internal Evaluation Marks (CIE):**

Attendance	Assignment/ Microproject	Internal Examination-1 (Written)	Internal Examination- 2 (Written )	Total
5	15	10	10	40

#### **End Semester Examination Marks (ESE)**

In Part A, all questions need to be answered and in Part B, each student can choose any one full question out of two questions

Part A	Part B	Total
• 2 Questions from each	Each question carries 9 marks.	
module.	Two questions will be given from each module, out	
• Total of 8 Questions, each	of which 1 question should be answered.	<b>60</b>
carrying 3 marks	Each question can have a maximum of 3 sub	60
	divisions.	
(8x3 =24marks)	(4x9 = 36  marks)	

## **Course Outcomes (COs)**

At the end of the course students should be able to:

		Bloom's
	Course Outcome	Knowledge
		Level (K)
CO1	Outline the basic bioelectric potentials and their implications in	K2
COI	diagnostics	
CO2	Summarize the principles used for diagnosis of abnormalities in the	K2
CO2	cardiovascular system	
CO3	Identify the techniques used for diagnosis and therapy in the	K2
003	neuromuscular and myoelectric systems.	
CO4	Illustrate the principle and working of different types of bio medical	K2
004	equipment/devices	
CO5	State various diagnostic medical imaging techniques.	K2

Note: K1- Remember, K2- Understand, K3- Apply, K4- Analyse, K5- Evaluate, K6- Create

	PO1	PO2	PO3	PO4	PO5	PO6	PO7	PO8	PO9	PO10	PO11	PO12
CO1	3					2						2
CO2	3					2						2
CO3	3					2	2					2
CO4	3					2	2					2
CO5	3					2	2					2

	Text Books						
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year			
1	Handbook of Biomedical Instrumentation	R. S. Khandpur	Tata Mc Graw Hill	Third edition			
2	Biomedical Instrumentation and Measurement	Leslie Cromwell, Fred J.  Weibell, Erich A.  Pfeiffer,	PHI	2nd Edition, 2004			

	Reference Books							
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year				
1	Medical Instrumentation application and design	John G Webster	John Wiley	5 <sup>th</sup> edition 2020				
2	Introduction to Biomedical Equipment Technology	J. J. Carr	Pearson Education	4 <sup>th</sup> edition 2020				
3	Principle of Biomedical Instrumentation and Measurement	Richard Aston	Merrill Education/Prentice Hall	1 <sup>st</sup> edition 2007				
4	Introduction to Biomedical Instrumentation	Barbara Christe	Cambridge University Press,	2 <sup>nd</sup> edition 2017				

	Video Links (NPTEL, SWAYAM)						
Module No.	Link ID						
1	https://www.youtube.com/watch?v=_fD9gOqiBVE						
2	http://www.digimat.in/nptel/courses/video/127106134/L16.html						

## **DATA STRUCTURES**

Course Code	PEECT522	CIE Marks	40
Teaching Hours/Week (L: T:P: R)	3:0:0:0	ESE Marks	60
Credits	3	Exam Hours	2 Hr. 30 Min.
Prerequisites (if any)	None/ (Course code)	Course Type	Theory

## **Course Objectives:**

1. To familiarise with different data structures and the techniques involved.

Module No.	Syllabus Description	Contact Hours
1	Basic Concepts of Data Structures: Algorithms, Performance Analysis, Space Complexity, Time Complexity, Asymptotic Notations  Arrays: Linear Search and Binary Search, Stacks, Queues-Circular Queues,	9
2	Priority Queues, Double Ended Queues, Evaluation of Expressions  Linked List: Self-Referential Structures, Dynamic Memory Allocation, Singly Linked List- Operations on Linked List. Doubly Linked List, Circular Linked List, Stacks and Queues using Linked List, Polynomial representation using Linked List	9
3	Trees and Graphs: Trees, Binary Trees-Tree Operations, Binary Tree Representation, Tree Traversals, Binary Search Trees- Binary Search Tree Operations Graphs, Representation of Graphs, Depth First Search and Breadth First Search on Graphs, Applications of Graphs	9
4	Sorting and Hashing: Sorting Techniques – Selection Sort, Insertion Sort, Quick Sort, Merge Sort and Heap Sort Hashing- Hashing Techniques, Collision Resolution, Overflow handling, Hashing functions – Mid square, Division, Folding, Digit Analysis	9

(CIE: 40 marks, ESE: 60 marks)

#### **Continuous Internal Evaluation Marks (CIE):**

Attendance	Assignment/ Microproject	Internal Examination-1 (Written)	Internal Examination- 2 (Written)	Total
5	15	10	10	40

## **End Semester Examination Marks (ESE)**

In Part A, all questions need to be answered and in Part B, each student can choose any one full question out of two questions

Part A	Part B			
• 2 Questions from each	Each question carries 9 marks.			
module.	Two questions will be given from each module, out			
• Total of 8 Questions, each	of which 1 question should be answered.	60		
carrying 3 marks	• Each question can have a maximum of 3 sub	00		
	divisions.			
(8x3 =24marks)	(4x9 = 36  marks)			

## Course Outcomes (COs)

At the end of the course students should be able to:

	Course Outcome	Bloom's Knowledge Level (KL)
CO1	Compare performance of algorithms using asymptotic notations	<b>K2</b>
CO2	Solve real world problems efficiently using appropriate data structures like arrays, linked list, stacks and queues.	К3
CO3	Make use of nonlinear data structures like trees and graphs to design algorithms for various applications.	К3
CO4	Apply and compare various techniques for searching and sorting.	К3
CO5	Apply appropriate hash function to store and access a given dataset	К3

Note: K1- Remember, K2- Understand, K3- Apply, K4- Analyse, K5- Evaluate, K6- Create

## **CO-PO Mapping Table (Mapping od Course Outcomes to Program Outcomes)**

	PO1	PO2	PO3	PO4	PO5	PO6	PO7	PO8	PO9	PO10	PO11	PO12
CO1	2	2	2	1	-	1	-	-	-	-	-	-
CO2	3	2	3	1	-	1	-	-	-	-	-	-
CO3	3	2	3	1	-	1	-	-	-	-	-	-
CO4	2	2	3	1	-	1	-	-	-	-	-	-
CO5	3	2	2	1	-	1	-	-	-	-	-	-

Note: 1: Slight (Low), 2: Moderate (Medium), 3: Substantial (High), -: No Correlation

	Text Books							
Sl. No	Title of the Book	Name of the	Name of the	Edition				
	Title of the book	Author/s	Publisher	and Year				
	Fundamentals of Data Structures	Ellis Horowitz, Sartaj						
1	in C	Sahni and Susan	Universities Press	2/e, 2008				
	III C	Anderson-Freed						
2	Classic Data Structures	Samanta D	Prentice Hall India	2/e, 2009				

	Reference Books							
Sl. No	Title of the Book Name of the Author/s		Name of the Publisher	Edition and Year				
1	Data Structures: A Pseudocode Approach with C	Richard F. Gilberg, Behrouz A. Forouzan	Cengage Learning	2/e, 2005				
2	Data Structures and Algorithms	Aho A. V., J. E. Hopcroft and J. D. Ullman	Pearson Publication	1/e. 1983				
3	Introduction to Data Structures with Applications	Tremblay J. P. and P. G. Sorenson	Tata McGraw Hill	2/e, 1995				
4	Advanced Data Structures	Peter Brass	Cambridge University Press	2/e, 2018				
5	Theory and Problems of Data Structures	Lipschuts S.	Schaum's Series	2/e, 2016				

	Video Links (NPTEL, SWAYAM)				
Module No.	Link ID				
1	https://nptel.ac.in/courses/106102064 https://youtu.be/zWg7U0OEAoE https://youtu.be/g1USSZVWDsY https://youtu.be/PGWZUgzDMYI				
2	https://nptel.ac.in/courses/106102064 https://youtu.be/PGWZUgzDMYI				
3	https://nptel.ac.in/courses/106102064 https://youtu.be/tORLeHHtazM https://youtu.be/eWeqqVpgNPg https://youtu.be/9zpSs845wf8				
4	https://youtu.be/KW0UvOW0XIo https://youtu.be/gtWw_8VvHjk				

## **SENSORS AND ACTUATORS**

Course Code	PEECT 523	CIE Marks	40
Teaching Hours/Week (L: T:P: R)	3:0:0:0	ESE Marks	60
Credits	3	Exam Hours	2 Hrs. 30 min.
Prerequisites (if any)	None	Course Type	Theory

## **Course Objectives:**

1. The course on Sensors and Actuators typically aims to provide students with comprehensive knowledge in the principles, design, and application of various sensors and actuators used in real-world applications

Module	Syllabus Description	Contact	
No.	Synabus Description		
	Introduction to Sensors and actuators: Block diagram of a closed loop		
	control System, Sensors and Transducers, Sensors Classification, Sensor		
1	Characteristics - Transfer Function, Calibration, Span (Full Scale Input),		
	Full-Scale Output, Accuracy, Precision, Hysteresis, Nonlinearity, Saturation,	9	
	Repeatability, Dead Band, Sensitivity, Resolution.		
	Position and Displacement Sensors - Potentiometric Sensors, Capacitive		
	Sensors, LVDT, Hall Effect Sensors		
	Pressure Sensors -Mercury Pressure Sensor, Bellows, Membranes, and Thin		
2	plates, Piezoresistive Sensors, Capacitive Sensors.		
	Force, Strain, and Tactile Sensors - Strain Gauges, Tactile Sensors - Switch	9	
	Sensors, Piezoelectric Sensors, Piezoresistive Sensors, Capacitive Touch		
	Sensors, Acoustic Touch Sensors, Optical Touch Sensors, Piezoelectric		
	Force Sensors.		
	Flow Sensors - Ultrasonic Flow Sensors, Electromagnetic Flow Sensors.		
3	Temperature Sensors - Resistance Temperature Detectors, Thermistors,	9	
	Thermocouple.		

	Proximity Sensors - PIR sensors. Ultrasonic proximity sensors.					
	Smart Sensors - Block Diagram, Difference between Normal Sensor &					
	Smart Sensor, Advantages, Disadvantages and Applications.					
4	Actuators: - Definition- classification-Electric, Hydraulic, Pneumatic actuators.  Hydraulic System - Physical Components and typical circuit. Hydraulic actuators - Linear actuators, Rotary actuators - Gear motor, vane motor.  Pneumatic System - Components and typical circuit. Pneumatic Actuators - Bellows actuator, Flapper-nozzle, Diaphragm actuators for industrial control valves.  Electric actuators- Solenoids, Stepper motors, DC motors, DC servo motors.	9				
	Electro-Pneumatic actuator; rotary output actuators, Linear output actuators.					

(CIE: 40 marks, ESE: 60 marks)

#### **Continuous Internal Evaluation Marks (CIE):**

Attendance	Assignment/ Microproject	Internal Examination-1 (Written)	Internal Examination- 2 (Written)	Total
5	15	10	10	40

## **End Semester Examination Marks (ESE)**

Part A	Part B	Total
2 Questions from each	Each question carries 9 marks.	
module.	Two questions will be given from each module, out	
• Total of 8 Questions, each	of which 1 question should be answered.	60
carrying 3 marks	• Each question can have a maximum of 3 sub	60
	divisions.	
(8x3 =24 marks)	(4x9 = 36  marks)	

At the end of the course students should be able to:

	Course Outcome	Bloom's Knowledge Level (KL)
CO1	Describe Sensor Fundamentals	К2
CO2	Explain the basic principles and concepts of commonly used different types of sensors, including their purpose, how they work, and the various types of sensors available.	K2
CO3	Illustrate the working principles of smart sensors	K2
CO4	Explain the working principle of different types of actuators.	К2

Note: K1- Remember, K2- Understand, K3- Apply, K4- Analyse, K5- Evaluate, K6- Create

	PO1	PO2	PO3	PO4	PO5	PO6	PO7	PO8	PO9	PO10	PO11	PO12
CO1	2	2	3			2						2
CO2	3	2	2			2						2
CO3	2	2	2			2						2
CO4	3	2	3			2						2

	Text Books								
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year					
1	Handbook of Modern Sensors	Jacob Fraden	Springer	Fourth Edition, 2010					
2	Hydraulics and Pneumatics	Andrew Parr	Elsevier Science	Second edition, 1999					
3	Process Control	K. Krishnaswamy	New Age International	Second edition, 2009					

	Reference Books							
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year				
1	Sensors and Actuators in Mechatronics, Design and Applications	Andrzej M. Pawlak	Taylor & Francis Group	1/e, 2016				
2	Mechatronic systems, Sensors and Actuators Fundamentals and Modelling	Robert H. Bishop	Taylor & Francis Group	3/e, 2022				
3	Process Control Instrumentation Technology	Curtis D. Johnson	Pearson/Prentice Hall	8/e, 2019				
4	Sensors and Transducers	D. Patranabis	PHI Learning	4/e, 2021				

Video Links (NPTEL, SWAYAM)							
Module No.	Link ID						
1	https://onlinecourses.nptel.ac.in/noc21_ee32/preview						
2	https://onlinecourses.nptel.ac.in/noc21_ee32/preview						
3	https://onlinecourses.nptel.ac.in/noc21_ee32/preview						
4	https://onlinecourses.nptel.ac.in/noc21_ee32/preview						

SEMESTER S5
ARM ARCHITECTURE AND PROGRAMMING

Course Code	PEECT524	CIE Marks	40
Teaching Hours/Week (L: T:P: R)	3:0:0:0	ESE Marks	60
Credits	3	Exam Hours	2 Hrs. 30 Min.
Prerequisites (if any)	Programming in C	Course Type	Theory

## **Course Objectives:**

- 1. To introduce ARM Cortex M programming in assembly and C
- 2. To lay the foundation for practical embedded system design

Module	Syllabus Description				
No.	Synabus Description				
	Embedded C: Fixed-width integer data types in C99, boolean type, mixing				
	types, manipulating bits in memory and IO ports, accessing memory mapped				
	IO using pointers, structures, packed structures, bit fields, casting address of				
	an object, unions.				
1	[1] Ch. 4				
1	Review of computer organization: Memory, CPU, IO, Introduction to Arm	9			
	cortex M architecture: Internal organization-general purpose and special				
	registers, instruction pipelining, memory model, bit banding, Arm assembly				
	language instruction format and operands				
	[1] Ch. 5				
	Arm assembly language programming: Loading constants into registers,				
	loading memory data into registers, storing data from registers to memory,				
	converting C assignment statements to assembly, memory address				
2	calculations, Memory addressing examples: translating C pointer expressions				
	to assembly, translating C subscript expressions to assembly, translating	9			
	structure references to assembly, Stack instructions, data processing				
	instructions: updating flags in APSR, arithmetic instructions, bit				

	manipulation instructions, shift instructions, bit field manipulation instructions	
	[1] Ch. 6	
3	Control structures in assembly language: instruction sequencing, conditional branch instructions, translating if-then and if-then-else statements to assembly, compound conditionals, implementing loops, speeding up array access, Implementing functions: function call and return, register usage, parameter passing, return values, temporary variables, preserving registers [1] Ch. 7.	9
4	IO programming in assembly: Interrupts and exceptions, thread and handler modes, entering the exception handler, returning from exception handler, latency reduction techniques, priorities and nested exceptions, synchronization, transfer rate and latency, buffers and queues, double buffering, polled waiting loops, interrupt driven IO, DMA [1] Ch. 8.  System initialization: Memory layout, cpu and vector table, C run-time environment, System Timer [1] Ch. 13	9

(CIE: 40 marks, ESE: 60 marks)

## **Continuous Internal Evaluation Marks (CIE):**

Attendance	Assignment/ Microproject	Internal Examination-1 (Written)	Internal Examination- 2 (Written)	Total
5	15	10	10	40

## **End Semester Examination Marks (ESE)**

Part A	Part B	Total
• 2 Questions from each	Each question carries 9 marks.	
module.	Two questions will be given from each module, out	
• Total of 8 Questions, each	of which 1 question should be answered.	(0
carrying 3 marks	Each question can have a maximum of 3 sub	60
	divisions.	
(8x3 =24 marks)	(4x9 = 36  marks)	

At the end of the course students should be able to:

	Course Outcome	Bloom's Knowledge Level (KL)
CO1	Use the features of C that are frequently used in embedded systems	К3
CO2	Explain a programmer's view of processor architecture	K2
CO3	Choose between programming at the level of assembly or C as appropriate	К3

Note: K1- Remember, K2- Understand, K3- Apply, K4- Analyse, K5- Evaluate, K6- Create

## **CO-PO Mapping Table (Mapping of Course Outcomes to Program Outcomes)**

	PO1	PO2	PO3	PO4	PO5	PO6	PO7	PO8	PO9	PO10	PO11	PO12
CO1	3	2	3	2								
CO2	3	2	3	2								
CO3	3	2	3	2								

Note: 1: Slight (Low), 2: Moderate (Medium), 3: Substantial (High), -: No Correlation

	Text Books						
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year			
1	Fundamentals of Embedded Software with the ARM Cortex M3	Daniel W Lewis	Pearson	2e, 2015			

	Reference Books						
Sl. No	Title of the Book	Title of the Book Name of the Author/s		Edition and Year			
1	The Definitive Guide to ARM Cortex-M3 and Cortex-M4 Processors	Joseph Yiu	Elsevier	3e, 2014			
2	Embedded systems with ARM Cortex M Microcontrollers in Assembly and C	Yifeng Zhu	E-man Press	3e, 2018			

	Video Links (NPTEL, SWAYAM)					
Module No.	Link ID					
1	https://archive.nptel.ac.in/courses/117/106/117106111/					
2	https://archive.nptel.ac.in/courses/106/105/106105193/					
3	https://onlinecourses.nptel.ac.in/noc22_cs93/preview					

### **SEMESTER S5**

## HIGH SPEED DIGITAL DESIGN

Course Code	PEECT526	CIE Marks	40
Teaching Hours/Week (L: T:P: R)	3:0:0:0	ESE Marks	60
Credits	3	Exam Hours	2 Hrs. 30 Min.
Prerequisites (if any)	BASIC ELECTRONICS	Course Type	Theory

## **Course Objectives:**

- 1. To understand the fundamentals of the effects of passive circuit elements on signal propagation in high speed digital circuits
- 2. To understand the high speed properties of logic gates and the measurement techniques at high frequencies
- **3.** To analyse the effects of wiring, source, and load on the signal propagation from one end of a circuit to the other end
- 4. To design the power supply and clock distribution circuits for high speed devices,

#### **SYLLABUS**

Module No.	Syllabus Description			
1	High Speed Digital Design: Fundamentals: Frequency and time, Time and distance, Lumped versus distributed systems, four kinds of reactance-ordinary capacitance and inductance, mutual capacitance and inductance, Relation of mutual capacitance and mutual inductance to cross talk.	9		
2	High Speed properties of Logic gates: Power, Quiescent vs active dissipation, Active power driving a capacitive load, Active power due to overlapping bias currents, Input power, Speed, Packaging (Power dissipation analysis of only CMOS logic gates are required)  Measurement Techniques: Rise time and bandwidth of oscilloscope probes, self inductance of probe ground loop, spurious signal pick up from probe ground loops, special probing fixtures, Avoiding pickup from probe shield currents, slowing down of a system clock, observing metastable states.	9		

		Transmission Lines: Problems of point to point wiring, signal distortion,					
		EMI, cross talk. Infinite Uniform transmission line; ideal distortion less					
	2	lossless transmission line, RC transmission line, Skin effect, Proximity	9				
	3	effect, Dielectric loss. Effects of source and load impedance. Termination:	9				
		End terminator, Source terminators, middle terminators, AC biasing for end					
		terminators, Resistor selection, Cross talk in terminators.					
		Power system: Stable voltage reference, Uniform voltage distribution,					
		distribution problems, choosing a bypass capacitor.					
	4	Clock Distribution: Timing margin, Clock skew, Using low impedance					
	4	drivers, using low impedance distribution lines, delay adjustments,	9				
		Differential distribution, Clock signal duty cycle, Decoupling clock receivers					
		from the clock bus. Clock Oscillators, Canned clock oscillator, Clock Jitter.					
1							

#### **Course Assessment Method**

(CIE: 40 marks, ESE: 60 marks)

### **Continuous Internal Evaluation Marks (CIE):**

Attendance	Assignment/ Microproject	Internal Examination-1 (Written)	Internal Examination- 2 (Written)	Total
5	15	10	10	40

# **End Semester Examination Marks (ESE)**

In Part A, all questions need to be answered and in Part B, each student can choose any one full question out of two questions

Part A	Part B	Total
2 Questions from each	Each question carries 9 marks.	
module.	Two questions will be given from each module, out	
• Total of 8 Questions, each	Total of 8 Questions, each of which 1 question should be answered.	
carrying 3 marks	• Each question can have a maximum of 3 sub	60
	divisions.	
(8x3 =24 marks)	(4x9 = 36  marks)	

## **Course Outcomes (COs)**

At the end of the course students should be able to:

	Course Outcome	Bloom's Knowledge Level (KL)
CO1	Illustrate the fundamentals of the effects of passive circuit elements on signal propagation in high speed digital circuits	К2
CO2	Describe the high speed properties of logic gates and the measurement techniques at high frequencies	K2
CO3	Analyze the effects of wiring, source, and load on the signal propagation from one end of a circuit to the other end	К3
CO4	Design the power supply and clock distribution circuits for high speed devices	К3

Note: K1- Remember, K2- Understand, K3- Apply, K4- Analyse, K5- Evaluate, K6- Create

# **CO-PO Mapping Table (Mapping of Course Outcomes to Program Outcomes)**

	PO1	PO2	PO3	PO4	PO5	PO6	PO7	PO8	PO9	PO10	PO11	PO12
CO1	3	3	3									3
CO2	3	3	3									3
CO3	3	3	3									3
CO4	3	3	3									3

Note: 1: Slight (Low), 2: Moderate (Medium), 3: Substantial (High), -: No Correlation

Text Books							
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year			
1	High Speed Digital Design: A Handbook of Black Magic	Howard Johnson & Martin Graham	Prentice Hall PTR,	Second Edition, 2008			
2	Noise Reduction Techniques in Electronic Systems	Henry W. Ott	John Wiley & Sons	Second Edition, 1988			

	Reference Books						
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year			
1	High-Speed Digital System Design—A Handbook of Interconnect Theory and Design Practices	Stephen H. Hall Garrett W. Hall James A. McCall	John Wiley & Sons	First Edition, 2000			
2	Digital Systems Engineering	William S. Dally & John W. Poulton	Cambridge University Press,	First Edition, 1998			
3	High Speed Digital Circuits	Masakazu Shoji	Addison Wesley Publishing Company	First Edition, 1996			
4	Digital Integrated Circuits: A Design perspective,	Jan M, Rabaey	Pearson	Second Edition, 2003			

	Video Links (NPTEL, SWAYAM)
Module	I interp
No.	Link ID
1	https://nptel.ac.in/courses/108105375
2	https://nptel.ac.in/courses/108105375
3	https://nptel.ac.in/courses/108105375, https://nptel.ac.in/courses/108106069
4	https://nptel.ac.in/courses/108105375

## **SEMESTER S5**

# **ESTIMATION AND DETECTION**

Course Code	PEECT527	CIE Marks	40
Teaching Hours/Week (L: T:P: R)	3:0:0:0	ESE Marks	60
Credits	3	Exam Hours	2 Hrs. 30 Min.
Prerequisites (if any)	None/ (Course code)	Course Type	Theory

# **Course Objectives:**

**1.** This course aims to impart the fundamentals of statistical signal processing theory in engineering applications.

# **SYLLABUS**

Module	Syllabus Description	Contact
No.	Synabus Description	Hours
1	Statistical Estimation Theory I  Fundamentals of estimation theory, the mathematical estimation problem,  Minimum variance unbiased estimation, basics of Cramer-Rao Lower  Bound, linear models, best linear unbiased estimation, application examples.	9
2	Statistical Estimation Theory II  Maximum likelihood estimation, least squares, Bayesian philosophy, minimum mean square error estimation, application examples.	9
3	Statistical Detection Theory I  Fundamentals of detection theory, the mathematical detection problem, Hypothesis testing, classical approach, Neyman-Pearson theorem, likelihood ratio test, receiver operating characteristics, Bayesian approach, minimum probability of error, Bayes risk, multiple hypothesis testing.	10
4	Statistical Detection Theory II  Detection of deterministic signals, matched filters, detection of random signals, estimator-correlator, linear model, application examples.	8

#### **Course Assessment Method**

(CIE: 40 marks, ESE: 60 marks)

### **Continuous Internal Evaluation Marks (CIE):**

Attendance	Assignment/ Microproject	Internal Examination-1 (Written)	Internal Examination- 2 (Written)	Total
5	15	10	10	40

# **End Semester Examination Marks (ESE)**

In Part A, all questions need to be answered and in Part B, each student can choose any one full question out of two questions

Part A	Part B	Total
• 2 Questions from each	Each question carries 9 marks.	
module.	Two questions will be given from each module, out	
• Total of 8 Questions, each	of which 1 question should be answered.	(0
carrying 3 marks	• Each question can have a maximum of 3 sub	60
	divisions.	
(8x3 =24marks)	(4x9 = 36  marks)	

## **Course Outcomes (COs)**

At the end of the course students should be able to:

	Course Outcome	Bloom's Knowledge Level (KL)
CO1	Summarize the fundamentals of statistical estimation principles used in various engineering problems.	К2
CO2	Apply different types of estimation algorithms in engineering applications.	К3
CO3	Illustrate the fundamentals of statistical detection principles used in various engineering problems.	К2
CO4	Apply various types of statistical decision rules in engineering applications.	К3

Note: K1- Remember, K2- Understand, K3- Apply, K4- Analyse, K5- Evaluate, K6- Create

# **CO-PO Mapping Table:**

	PO1	PO2	PO3	PO4	PO5	PO6	PO7	PO8	PO9	PO10	PO11	PO12
CO1	3											2
CO2	3	3	3	3	3							2
CO3	3											2
CO4	3	3	3	3	3							2

	Text Books						
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year			
1	"Fundamentals of Statistical Signal Processing" Vol I: Estimation Theory,	S.M. Kay,	Pearson,	3/e, 2010.			
2	"Fundamentals of Statistical Signal Processing" Vol II: Detection Theory,	S.M. Kay,	Pearson,	3/e, 2010.			

	Reference Books						
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year			
1	Detection, Estimation, and Modulation Theory, Vol. I,	H. L. Van Trees	John Wiley & Sons	2/e, 2001			
2	Statistical Digital Signal Processing and Modelling	Monson H. Hayes	John Wiley & Sons	2/e, 2018			

	Video Links (NPTEL, SWAYAM)						
Module No.	Link ID						
1	https://nptel.ac.in/courses/117103018						
2	https://nptel.ac.in/courses/117103018						
3	https://nptel.ac.in/courses/117103018						
4	https://nptel.ac.in/courses/117103018						

SEMESTER S5

# ARM ARCHITECTURE, PROGRAMMING AND INTERFACING

Course Code	PEECT525	CIE Marks	40
Teaching Hours/Week (L: T:P: R)	3:0:0:0	ESE Marks	60
Credits	5/3	Exam Hours	2 Hrs.30 Min.
Prerequisites (if any)	None	Course Type	Theory

# **Course Objectives:**

- 1. To introduce ARM Cortex M programming in assembly and C
- 2. To lay the foundation for practical embedded system design

# **SYLLABUS**

Module	Syllabus Description	Contact
No.	Synabus Description	Hours
1	Embedded C: Fixed-width integer data types in C99, boolean type, mixing types, manipulating bits in memory and IO ports, accessing memory mapped IO using pointers, structures, packed structures, bit fields, casting address of an object, unions.  [1] Ch. 4  Review of computer organization: Memory, CPU, IO, Introduction to Arm cortex M architecture: Internal organization-general purpose and special registers, instruction pipelining, memory model, bit banding, Arm assembly language instruction format and operands  [1] Ch. 5	9
2	Arm assembly language programming: Loading constants into registers, loading memory data into registers, storing data from registers to memory, converting C assignment statements to assembly, memory address calculations, Memory addressing examples: translating C pointer expressions to assembly, translating C subscript expressions to assembly, translating structure references to assembly, Stack instructions, data processing	9

	instructions: updating flags in APSR, arithmetic instructions, bit	
	manipulation instructions, shift instructions, bit field manipulation	
	instructions	
	[1] Ch. 6	
	Control structures in assembly language: instruction sequencing, conditional	
	branch instructions, translating if-then and if-then-else statements to	
3	assembly, compound conditionals, implementing loops, speeding up array	9
3	access, Implementing functions: function call and return, register usage,	,
	parameter passing, return values, temporary variables, preserving registers	
	[1] Ch. 7.	
	IO programming in assembly: Interrupts and exceptions, thread and handler	
	modes, entering the exception handler, returning from exception handler,	
	latency reduction techniques, priorities and nested exceptions,	
	synchronization, transfer rate and latency, buffers and queues, double	
4	buffering, polled waiting loops, interrupt driven IO, DMA	9
	[1] Ch. 8.	
	System initialization: Memory layout, cpu and vector table, C run-time	
	environment, System Timer	
	[1] Ch. 13	

### **Course Assessment Method**

(CIE: 40 marks, ESE: 60 marks)

### **Continuous Internal Evaluation Marks (CIE):**

Attendance	Internal Ex	Evaluate	Analyse	Total
5	15	10	10	40

## Criteria for Evaluation (Evaluate and Analyse): 20 marks

Interfacing experiments on Arm Microcontroller boards TM4C123G/STM32 Nucleo

GPIO – push button, LED, keypad scan ([2] Ch. 14)

Toggling LED using timers ([2] Ch. 15)

Stepper motor control ([2] Ch. 16)

LCD interfacing ([2] Ch. 17)

ADC and DAC with DMA ([2] Ch. 19, 20, 21)

Serial Communication ([2] Ch. 22)

Course Project involving the design and implementation of an embedded system for a chosen application

Project phases: Proposal, Implementation, Testing, Final Report, Presentations and Viva Voce

### **End Semester Examination Marks (ESE):**

In Part A, all questions need to be answered and in Part B, each student can choose any one full question out of two questions

Part A	Part B	Total
2 Questions from each	2 questions will be given from each module, out of	
module.	which 1 question should be answered. Each question	
• Total of 8 Questions,	can have a maximum of 3 sub divisions. Each	60
each carrying 3 marks	question carries 9 marks.	
(8x3 =24marks)	(4x9 = 36  marks)	

### **Course Outcomes (COs)**

At the end of the course students should be able to:

	Course Outcome	Bloom's Knowledge Level (KL)
CO1	Use the features of C that are frequently used in embedded systems	К3
CO2	Explain a programmer's view of processor architecture	K2
CO3	Choose between programming at the level of assembly or C as appropriate	К3
CO4	Analyze the interfacing of peripherals	K4

Note: K1- Remember, K2- Understand, K3- Apply, K4- Analyse, K5- Evaluate, K6- Create

# **CO-PO Mapping Table (Mapping of Course Outcomes to Program Outcomes)**

	PO1	PO2	PO3	PO4	PO5	PO6	PO7	PO8	PO9	PO10	PO11	PO12
CO1	3											2
CO2	3		3	3								2
CO3	3		3	3								2
CO4	3		3	3								2

Note: 1: Slight (Low), 2: Moderate (Medium), 3: Substantial (High), -: No Correlation

	Text Books							
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year				
1	Fundamentals of Embedded Software with the ARM Cortex M3	Daniel W Lewis	Pearson	2e, 2015				
2	Embedded systems with ARM Cortex M Microcontrollers in Assembly and C	Yifeng Zhu	E-man Press	3e, 2018				

	Reference Books						
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year			
1	The Definitive Guide to ARM Cortex-M3 and Cortex-M4 Processors	Joseph Yiu	Elsevier	3e, 2014			

	Video Links (NPTEL, SWAYAM)					
Module No.	Link ID					
1	Modern Embedded Systems Programming Course Quantum Leaps, LLC https://youtube.com/playlist?list=PLPW8O6W-1chwyTzI3BHwBLbGQoPFxPAPM&si=vmU66G3vMmQihUPk					
2	Modern Embedded Systems Programming Course Quantum Leaps, LLC https://youtube.com/playlist?list=PLPW8O6W- 1chwyTzI3BHwBLbGQoPFxPAPM&si=vmU66G3vMmQihUPk					
3	Modern Embedded Systems Programming Course Quantum Leaps, LLC https://youtube.com/playlist?list=PLPW8O6W- 1chwyTzI3BHwBLbGQoPFxPAPM&si=vmU66G3vMmQihUPk					
4	Modern Embedded Systems Programming Course Quantum Leaps, LLC https://youtube.com/playlist?list=PLPW8O6W-1chwyTzI3BHwBLbGQoPFxPAPM&si=vmU66G3vMmQihUPk					

## **SEMESTER S5**

# DIGITAL SIGNAL PROCESSING LAB

Course Code	PBECL507	CIE Marks	50
Teaching Hours/Week	0:0:3:0	ESE Marks	50
(L: T:P: R)			
Credits	2	Exam Hours	2 Hrs. 30 Min.
Prerequisites (if any)	Signals & Systems, DSP	Course Type	Lab

# **Course Objectives:**

- 1. To realize the DFT, filtering techniques and familiarize DSP hardware
- 2. To implement Digital Filter.

# **Details of Experiment**

Expt.	Experiment
	Simulation of Signals Simulate the following signals using Python/
1	Scilab/MATLAB.
1	1. Unit impulse signal 2. Unit pulse signal 3. Unit ramp signal 4. Bipolar pulse 5. Triangular
	signal
	Verification of the Properties of DFT
	1. Generate a DFT matrix and apply it to an example sequence.
	2. Write a function that returns the N point DFT matrix VN for a given N.
	3. Plot its real and imaginary parts of VN as images using matshow or imshow commands (in
	Python) for $N = 16$ , $N = 64$ and $N = 1024$
	4. Compute the DFTs of 16 point, 64 point and 1024 point random sequences using the above
2	matrices.
	5. Observe the time of computations for $N = 2\sigma$ for different values of $\sigma$ . (You may use
	the time module in Python).
	6. Use some iterations to plot the times of computation against x. Plot and understand this
	curve. Plot the computation times for the FFT function over this curve and observe the
	computational advantage of FFT.
	Circular Convolution.
	1. Write a python function <i>circcon.py</i> that returns the circular convolution of an N1 point

sequence and an N2 point sequence given at the input. The easiest way is to convert a linear convolution into circular convolution with N = max(N1, N2).  Parseval's Theorem  Take two complex random sequences x1[n] and x2[n], and verify Parseval's Theorem.  Familiarization of DSP Hardware  1. Familiarization of the code composer studio (in the case of T1 hard-ware) or Visual DSP (in the case of 'Analog Devices hardware) or any equivalent cross-compiler for DSP programming.  2. Familiarization of the analog and digital input and output ports of the DSP board.  3. Generation and cross compilation and execution of the C code to connect the input digital switches to the output LFDs.  4. Generation and cross compilation and execution of the C code to connect the input analog port to the output. Connect a microphone, speak into it and observe the output electrical signal on a DSO and store it.  Linear convolution  1. Write a C function for the linear convolution of two arrays.  2. The arrays may be kept in different files and downloaded to the DSP hardware.  3. Store the result as a file and observe the output.  FFT of signals  1. Write a C function for N - point FFT.  2. Connect a precision signal generator and apply 1 mV , 1 kHz sinusoid at the analog port.  3. Apply the FTT on the imput signal with appropriate window size and observe the result.  4. Connect microphone to the analog port and read in real time speech.  5. Observe and store the FFT values.  IFFT with FFT  1. Use the FFT function in the previous experiment to compute the IFFT of the input signal.  2. Apply IFFT on the stored FFT values from the previous experiments and observe the reconstruction.  FIR low pass filter  1. Use Python/scilab to implement the FIR filter response h[n] = ωc n)/π for a filter size N = 50, ωc = 0.1π and ωc = 0.3π.  2. Realize the harming (wH [n]) and kaiser (wK[n]) windows.  3. Compute h[n]w[n] in both cases and store as file.  4. Observe the low pass response in the simulator.  5. Download the filter on to the DSP target bo					
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	6. Test the operation of the filters with speech signals.		
	1 2		
	Overlap Save Block Convolution		
	1. Use the file of filter coefficients from the previous experiment.		
	2. Realize the system shown below for the input speech signal $x[n]$ .		
8	3. Segment the signal values into blocks of length $N = 2000$ . Pad the last block with zeros, if necessary.  4. Implement the <i>overlap save</i> block convolution method		
	9. Overlap Add Block Convolution		
	1. Use the file of filter coefficients from the previous experiment.		
9	2. Realize the system shown in the previous experiment for the input speech signal $x[n]$ .		
	3. Segment the signal values into blocks of length $N = 2000$ . Pad the last block with zeros, if		
	necessary.		
	4. Implement the <i>overlap add</i> block convolution method.		

# Course Assessment Method (CIE: 50 Marks, ESE 50 Marks)

# **Continuous Internal Evaluation Marks (CIE):**

Attendance	Preparation/Pre-Lab Work, experiments, Viva and Timely completion of Lab Reports / Record.  (Continuous Assessment)	Internal Exam	Total
5	25	20	50

# **End Semester Examination Marks (ESE):**

Procedure/ Preparatory work/Design/ Algorithm	Conduct of experiment/ Execution of work/ troubleshooting/ Programming	Result with valid inference/ Quality of Output	Viva voce	Record	Total
10	15	10	10	5	50

## **Mandatory requirements for ESE:**

- Submission of Record: Students shall be allowed for the end semester examination only upon submitting the duly certified record.
- Endorsement by External Examiner: The external examiner shall endorse the record.

### **Course Outcomes (COs)**

At the end of the course the student will be able to:

	Course Outcome	Bloom's Knowledge Level (KL)
CO1	Generate basic signal waveforms	K2
CO2	Verify the properties of DFT	К2
CO3	Familiarize with DSP hardware and interface with Computer	K2
CO4	Implement LTI systems	К3
CO5	Design and Implement FIR low-pass filters	К3

K1- Remember, K2- Understand, K3- Apply, K4- Analyse, K5- Evaluate, K6- Create

# **CO-PO Mapping Table**

	PO1	PO2	PO3	PO4	PO5	PO6	PO7	PO8	PO9	PO10	PO11	PO12
CO1	3	3	1	2	3							1
CO2	3	3	1	3	3							1
CO3	3	2	3	3	3							1
CO4	3	3	2	3	3							1
CO5	3	3	3	2	3							1

1: Slight (Low), 2: Moderate (Medium), 3: Substantial (High), : No Correlation

	Text Books									
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year						
1	Digital Signal Processing using Matlab	Vinay K. Ingle, John G. Proakis	Cengage Learning	3 <sup>rd</sup> Ed., 2011						
2	Think DSP: Digital Signal Processing using Python	Allen B. Downey	Green Tea Press	1 <sup>st</sup> Ed. 2019						
3	DSP applications using C and the TMS320C6x DSK	Chassaing, Rulph	Wiley & Sons	2/e. 2008						

Reference Books								
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year				
1	Discrete-Time Signal Processing	Alan V Oppenheim, Ronald W. Schafer	Pearson Education	4 <sup>th</sup> Ed.,2018				

	Video Links (NPTEL, SWAYAM)						
Sl. No.	Link ID						
1	https://www.youtube.com/watch?v=6dFnpz_AEyA						
2	https://onlinecourses.nptel.ac.in/noc21_ee20/preview						

# **Continuous Assessment (25 Marks)**

# 1. Preparation and Pre-Lab Work (7 Marks)

- Pre-Lab Assignments: Assessment of pre-lab assignments or quizzes that test understanding of the upcoming experiment.
- Understanding of Theory: Evaluation based on students' preparation and understanding of the theoretical background related to the experiments.

#### 2. Conduct of Experiments (7 Marks)

- Procedure and Execution: Adherence to correct procedures, accurate execution of experiments, and following safety protocols.
- Skill Proficiency: Proficiency in handling equipment, accuracy in observations, and troubleshooting skills during the experiments.
- Teamwork: Collaboration and participation in group experiments.

#### 3. Lab Reports and Record Keeping (6 Marks)

- Quality of Reports: Clarity, completeness and accuracy of lab reports. Proper documentation of
  experiments, data analysis and conclusions.
- Timely Submission: Adhering to deadlines for submitting lab reports/rough record and maintaining a well-organized fair record.

#### 4. Viva Voce (5 Marks)

 Oral Examination: Ability to explain the experiment, results and underlying principles during a viva voce session.

Final Marks Averaging: The final marks for preparation, conduct of experiments, viva, and record are the average of all the specified experiments in the syllabus.

#### **Evaluation Pattern for End Semester Examination (50 Marks)**

#### 1. Procedure/Preliminary Work/Design/Algorithm (10 Marks)

- Procedure Understanding and Description: Clarity in explaining the procedure and understanding each step involved.
- Preliminary Work and Planning: Thoroughness in planning and organizing materials/equipment.
- Algorithm Development: Correctness and efficiency of the algorithm related to the experiment.
- Creativity and logic in algorithm or experimental design.

#### 2. Conduct of Experiment/Execution of Work/Programming (15 Marks)

 Setup and Execution: Proper setup and accurate execution of the experiment or programming task.

## 3. Result with Valid Inference/Quality of Output (10 Marks)

- Accuracy of Results: Precision and correctness of the obtained results.
- Analysis and Interpretation: Validity of inferences drawn from the experiment or quality of program output.

### 4. Viva Voce (10 Marks)

- Ability to explain the experiment, procedure results and answer related questions
- Proficiency in answering questions related to theoretical and practical aspects of the subject.

### 5. Record (5 Marks)

• Completeness, clarity, and accuracy of the lab record submitted

## **SEMESTER S5**

# **COMMUNICATION LABI**

Course Code	PCECL508	CIE Marks	50
Teaching Hours/Week (L: T:P: R)	0:0:3:0	ESE Marks	50
Credits	2	Exam Hours	2 Hrs. 30 Min.
Prerequisites (if any)	Analog circuits, Signals and systems, Digital Signal Processing	Course Type	Lab

# **Course Objectives:**

- 1. Understanding and Implementing Modulation and Detection Techniques
- 2. Analyzing and Evaluating Communication System Performance.

### **Details of Experiment**

Expt. No	Experiment
	PART A: Hardware Experiments
	Any one from the following Analog modulation schemes
1	<ul> <li>AM modulation and detection using Transistors or ICS</li> <li>FM modulation and detection</li> </ul>
	Hardware Experiment: Any one from the following Digital modulation & Waveform coding Schemes
2	Generation and Detection of PCM signals
	Generation and Detection of Delta modulated signals
	Generation and Detection of BPSK
	Generation and Detection of QPSK
	PART B: Simulation Experiments
1.	Performance of Waveform Coding Using PCM

2.	Pulse Shaping and Matched Filtering
3.	Eye diagram
4.	Error Performance of BPSK
5.	Error Performance of QPSK
	PART C: Software Defined Radio
1.	Familiarization with Software Defined Radio (Hardware and Control Software)
2.	FM reception or FM transmission using SDR

## **Experiment Details**

#### **PART A: Hardware Experiments**

The students shall design and setup simple prototype circuits with the help of available ICs. They can observe waveforms produced by these circuits for standard ideal inputs

#### **PART B: Simulation Experiments**

The students shall write scripts to simulate components of communication systems for the following experiments.

#### Performance of Waveform Coding Using PCM

- 1. Generate a sinusoidal waveform with a DC offset so that it takes only
- 2. positive amplitude value.
- 3. Sample and quantize the signal using a uniform quantizer with number of
- 4. representation levels L. Vary L. Represent each value using decimal to
- 5. binary encoder.
- 6. Compute the signal-to-noise ratio in dB.
- 7. Plot the SNR versus number of bits per symbol. Observe that the SNR
- 8. increases linearly

### **Pulse Shaping and Matched Filtering**

- 1. Generate a string of message bits.
- 2. Use Root Raised Cosine (RRC) pulse p(t) as the shaping pulse, and generate the

- 3. corresponding baseband signal with a fixed bit duration Tb. You may use roll-off factor as  $\alpha = 0.4$ . Vary the roll off rate and study.
- 4. Simulate transmission of baseband signal via an AWGN channel
- 5. Apply matched filter with frequency response  $Pr(f) = P^*(f)$  to the received signal.
- 6. Sample the signal at mTb and compare it against the message sequence.

#### Eye diagram

- 1. Generate a string of message bits.
- 2. Use raised cosine pulse p(t) as the shaping pulse, and generate the corresponding baseband signal with a fixed bit duration Tb You may use roll-off factor as  $\alpha$ = 0.4
- Use various roll off factors and plot the eye diagram in each case for the received signal.Make a comparison study among them.

#### **Error Performance of BPSK**

- 1. Generate a string of message bits.
- 2. Encode using BPSK with energy per bit Eb and represent it using points in a signal-space.
- 3. Simulate transmission of the BPSK modulated signal via an AWGN channel with variance No/2.
- 4. Detect using an ML, decoder and plot the probability of error as a function of SNR per bit Eb/No.

#### **Error Performance of QPSK**

- 1. Generate a string of message bits.
- Encode using QPSK with energy per symbol E<sub>b</sub> and represent it using points in a signalspace.
- 3. Simulate transmission of the QPSK modulated signal via an AWGN channel with variance No/2 in both I-channel and Q-channel.
- 4. Detect using an ML decoder and plot the probability of error as a function of SNR per bit  $E_b/N_0$  where  $E_s$ =2 $E_b$

#### **PART C: Software Defined Radio**

The students shall emulate communication systems with the help of software-defined-radio hardware and necessary control software. Use available blocks in GNU Radio (or similar software's like Simulink/ Lab- View) to implement all the signal processing.

#### Familiarization with Software Defined Radio (Hardware and Control Software)

- 1. Familiarize with an SDR hardware for reception and transmission of RF signal
- 2. Familiarize how it can be interfaced with computer
- 3. Familiarize with GNU Radio (or similar software's like Simulink/ Lab- View) that can be used to process the signals received through the SDR hardware.
- 4. Familiarize available blocks in GNU radio. Study how signals can be generated and spectrum (or power spectral density) of signals can be analyzed. Study how filtering can be performed.

#### FM reception using SDR

- 1. Receive digitized FM signal (for the clearest channel in the lab) using the SDR board.
- 2. Set up an LPF and FM receiver using GNU Radio.
- 3. Use appropriate sink in GNU Radio to display the spectrum of signal.
- 4. Resample the voice to make it suitable for playing on computer speaker. or playing on compute

#### FM transmission using SDR

- 1. Use a wave file source.
- 2. Set up an FM transmitter using GNU Radio.
- 3. Resample the voice source and transmit using the SDR.

#### Course Assessment Method (CIE: 50 Marks, ESE 50 Marks)

### **Continuous Internal Evaluation Marks (CIE):**

Attendance	Preparation/Pre-Lab Work, experiments, Viva and Timely completion of Lab Reports / Record.  (Continuous Assessment)	Internal Exam	Total
5	25	20	50

# **End Semester Examination Marks (ESE):**

Procedure/ Preparatory work/Design/ Algorithm	Conduct of experiment/ Execution of work/ troubleshooting/ Programming	Result with valid inference/ Quality of Output	Viva voce	Record	Total
10	15	10	10	5	50

## **Mandatory requirements for ESE:**

- Submission of Record: Students shall be allowed for the end semester examination only upon submitting the duly certified record.
- Endorsement by External Examiner: The external examiner shall endorse the record.

# **Course Outcomes (COs)**

### At the end of the course the student will be able to:

	Course Outcome	Bloom's Knowledge Level (KL)
CO1	Setup simple prototype circuits for waveform coding and digital modulation techniques working in a team.	К3
CO2	Simulate the error performance of a digital communication system using standard binary and M-ary modulation schemes.	K4
CO3	Develop hands-on skills to emulate a communication system with software-designed-radio working in a team.	K5

K1- Remember, K2- Understand, K3- Apply, K4- Analyse, K5- Evaluate, K6- Create

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	Text Books								
Sl. No	Title of the Book	Title of the Book Name of the Author/s		Edition and Year					
1	The Hobbyist's Guide to the RTL-SDR: Really Cheap Software Defined Radio	Carl Laufer	CreateSpace Independent Publishing Platform	2 <sup>nd</sup> Edition, 2015					
2	Principles of Communication Systems Simulation with Wireless Applications	WH Tranter, KS Shanmugan, TS Rappaport, KL Kosbar	Prentice Hall	2 <sup>nd</sup> Edition, 2006					
3	Digital Modulations using Python	Mathuranathan Viswanathan, "	Independently Published	1 <sup>st</sup> Edition, 2019					

Reference Books					
Sl. No	Title of the Book	Name of the Author/s	Name of the Publisher	Edition and Year	
1	Communication Systems	Simon Haykin and Michael Moher	Wiley	5th Edition, 2020	
2	Modern Digital and Analog Communication Systems	B.P. Lathi and Zhi Ding	Oxford University Press	5th Edition, 2018	
3	Introduction to Analog and Digital Communication	Simon Haykin and Michael Moher	Wiley	2nd Edition, 2006	
4	Electronic communication systems	George Kennedy	McGraw Hil	6 <sup>th</sup> Edition, 2017	

Video Links (NPTEL, SWAYAM)			
Sl. No.	Link ID		
1	Neel Pandeya, "Implementation of a Simple FM Receiver in GNU Ra- dio," https://kb.ettus.com/		
2	Michael Ossmann, "Software Defined Radio with HackRF," YouTube Tutorial, https://www.youtube.com/watch?v=BeeSN14JUYU		
3	Nptel videos on Software Defined radio, https://www.youtube.com/watch?v=0KQWPFwFByU		
4	Experimenting with software defined radio, https://www.youtube.com/watch?v=tx5xofG2Fxg		

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