THE DEVELOPMENT OF DIGITAL NETWORKS

- 1.1 Introduction to Digital Telecommunications 1.1.1 The Digital Revolution 1.1.2 Digital Network Development 1.1.3 Pulse Code Modulation Fundamentals 1.1.4 Growth in Data Transmission
- 1.2 The Components of a Communication System 1.2.1 System Functions 1.2.2 Comparison of Analogue and Digital Systems
- 1.3 Digital Data Transmission 1.3.1 Voice band Data Transmission 1.3.2 Public Data Networks 1.3.3 Non-switched Data Networks 1.3.4 Packet Switched Data Networks 1.3.5 Local Area Networks
- 1.4 Integrated Services and Hierarchies 1.4.1 PCM Bit Rates and Hierarchies 1.4.2 Bit Rates for other Services
- 1.5 Scope of this text

CHAPTER 2

BASEBAND DIGITAL TRANSMISSION SIGNALS

- 2.1 Introduction 2.1.1 A Digital Transmission System 2.1.2 Baseband Signals
- 2.2 Baseband Line Transmission Systems
- 2.3 Algebraic Representation of Line Signals
- 2.4 Encoding and Pulse Shaping 2.4.1 System Elements 2.4.2 Alternate-mark-inversion (AMI) Code

2.5 Line Waveforms

- 2.6 Line Code Selection 2.6.1 Desirable Code Characteristics 2.6.2 AMI Code Properties 2.6.3 Manchester Code (twinned binary, split phase) 2.6.4 Differential Diphase Code
- 2.7 Methods for Calculating Frequency Spectra 2.7.1 Spectra of Periodic Signals 2.7.2 Spectra of Aperiodic Signals 2.7.3 Spectra of Random Waveforms
- 2.8 Power Spectral Density of Line Codes
- 2.9 Other Ternary Line Codes 2.9.1 High Density Bipolar (HDBn) Codes 2.9.2 4B3T Codes 2.9.3 5B4T, 7B5T, 8B6T, and 10B7T Codes

INTERSYMBOL ITERFERENCE AND PULSE SHAPING

- 3.1 Introduction
- 3.2 Nyquist Pulse Shaping 3.2.1 Maximum Rate Pulses 3.2.2 Symbol Packing Rate 3.2.3 Nyquist Vestigial Symmetry Criterion for Zero ISI 3.2.4 Raised Cosine Spectrum for Zero ISI 3.2.5 Pulse Shaping Circuits

3.3 Multilevel Signalling

3.4 Correlative (Partial Response) Signalling 3.4.1 Duobinary Scheme 3.4.2 Generalised Correlative (partial response) Encoding 3.4.3 Modified duobinary (Class 4 Partial Response) scheme

SIGNAL REGENERATION

- 4.1 Introduction
- 4.2 Regenerative Repeaters
 4.2.1 Functions
 4.2.2 Clock Recovery
 4.2.3 Sampling and Decision Circuits
- 4.3 Equalisers
 4.3.1 Functions
 4.3.2 Typical Equaliser Characteristics
 4.3.3 Transversal Equalisers
 4.3.4 Automatic Equalisers
 4.3.5 Computer Simulation
- 4.4 Bit-error Rate Calculations 4.4.1 Mathematical Models 4.4.2 Probability of error-wideband Gaussian noise case 4.4.3 Allocation of Transmit and Receive Gaussian Filtering

CHAPTER 5

MEASUREMENT TECHNIQUES

5.1 Introduction

5.2 Eye Diagrams 5.2.1 Measurement Procedures 5.2.2 Important Features of Eye Patterns 5.2.3 Effects of Intersymbol Interference 5.2.4 Effects of Noise and Crosstalk

- 5.3 Near End Crosstalk Noise Figure 5.3.1 Regenerator Performance Measurement 5.3.2 Input Signal to NEXT Noise Ratio
- 5.4 Pseudorandom Binary Test Signals 5.4.1 Introduction 5.4.2 Searching for a Random Sequence 5.4.3 Feedback Shift Register Generators 5.4.4 Properties of Pseudorandom Binary Signals 5.4.5 Applications of Pseudorandom Sequences

- 5.5 Error Rate Measurements 5.5.1 Bit-error Rates 5.5.2 Error-free Seconds
- 5.6 Regenerator Fault Location Tests 5.6.1 Triples Test Signal 5.6.2 Fault Location Procedure

DIGITAL RADIO SYSTEMS

- 6.1 Introduction 6.1.1 Historical Developments 6.1.2 Digital Versus Analogue Radio
- 6.2 Typical Digital Radio Systems
 6.2.1 Digital Radio Equipment
 6.2.2 Transmission Capacities and Frequency Bands
 6.2.3 Typical Equipment Characteristics
 6.2.4 Performance Objectives
- 6.3 Modulation Methods 6.3.1 Spectral Efficiency 6.3.2 Choice of Modulation Type 6.3.3 PSK (Phase-Shift-Keyed) Modulation Schemes 6.3.4 FSK (Frequency-Shift-Keying) Modulation 6.3.5 16 QAM (Quadrature-Amplitude) Modulation 6.3.6 Quadrature Partial Response Signalling (QPRS) 6.3.7 Minimum Shift Keying (MSK) Systems
- 6.4 Detection
 6.4.1 Optimum Detector for Binary PSK, FSK or ASK in Gaussian
 6.4.2 Coherent Detector for QAM and M-PSK Signals
 6.4.3 Optimum Detectors for Channels with ISI-Viterbi Algorithms
 6.4.4 Non-coherent Detectors
 6.4.5 Bit-error Rate Performance

6.5 Radio Link System Design

- 6.5.1 Free Space Calculations for Single Hops
- 6.5.2 Flat Fade Margin
- 6.5.3 Percentage Outage Prediction Vigant's Formula
- 6.5.4 Frequency Selective Fading Rummler's Model
- 6.5.5 Intersymbol Interference due to Frequency Selective Fading

6.5.6 Space Diversity 6.5.7 Adaptive Equalisation

- 6.6 Hybrid Radio Systems
 6.6.1 Data In Voice (DIV) Systems
 6.6.2 Data Above Voice (DAV) and Data Above Video (DAVID) Systems
 6.6.3 Data Over Voice (DOV) Systems
 6.6.4 Data Under Voice (DUV) Systems
- 6.7 Digital Concentrator Subscriber Radio Systems

CHAPTER 7

COMPUTER NETWORKS

- 7.1 Introduction
- 7.2 Classification of Computer Networks
- 7.3 Computer Network Structures

7.4 ISO Model for Open Systems Interconnection (OSI) 7.4.1 Physical Level Protocols 7.4.2 Data Link Layer 7.4.3 Error Checking in the BISYNC Protocol 7.4.4 Other Control Characters and Segments 7.4.5 Transparency 7.4.6 Multipoint Transmission 7.4.7 Synchronization 7.4.8 Byte Count Protocols 7.4.9 Bit-oriented Data Link Control Protocols

- 7.5 Network Layer 7.5.1 X.25 Recommendations 7.5.2 X.75 Recommendations
- 7.6 Higher Layers

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- 7.7 Multiple Access Techniques 7.7.1 Random Aloha 7.7.2 Slotted 7.7.3 Reservation
- 7.8 Local Area Networks
- 7.9 Network Design Fundamentals

ERROR CONTROL IN DIGITAL NETWORKS

- 8.1 Introduction
- 8.2 Errors and Erasures
- 8.3 Error Detection Using Block Codes 8.3.1 Single Bit Parity Detection 8.3.2 Weight Distribution of a Code 8.3.3 Error Detection Reliability of the Single-Parity Code 8.3.4 Linear Block Codes for Error Detection 8.3.5 Minimum Distance of a Code
- 8.4 Cyclic Codes for Error Detection 8.4.1 Polynomial Representation 8.4.2 Generator Polynomial 8.4.3 Generation of Parity (encoding) 8.4.4 Encoder for Rec. X.25 Frame Check Sequence 8.4.5 Decoding for Error Detection 8.4.6 Error Detection for the CCITT Rec. X.25 Code 8.4.7 Variable Block Lengths - Shortened Cyclic Codes 8.4.8 Probability of Un-Detected Error
- 8.5 Automatic-Repeat-Request (ARQ) Systems 8.5.1 ARQ Procedures 8.5.2 Throughput of Go-Back-N ARQ 8.5.3 Other ARQ Procedures
- 8.6 Hybrid ARQ Schemes
 8.6.1 Parity Retransmission ARQ Strategy
 8.6.2 Retransmission Protocols
 8.6.3 Choice of Error Correction Code
 8.6.4 Throughput Analysis

CHAPTER 9

DIGITAL SWITCHING

- 9.1 Digital Switching 9.1.1 Local Networks
 - 9.1.2 Concentrators
 - 9.1.3 Digital Group Selector
 - 9.1.4 Advantages of Concentrators with Centralised Exchanges
 - 9.1.5 Digital PCM Switching Techniques
 - 9.1.6 Exchange Congestion

- 9.2 Network Synchronization
 9.2.1 Synchronization Requirements Slips
 9.2.2 Causes of Slips
 9.2.3 Approaches to Network Synchronization
 9.2.4 Plesiochronous Networks
 9.2.5 Master-Slave Synchronization
 9.2.6 Mutual Synchronization
 9.2.7 Comparison of Synchronization Methods
 9.3 Frame Synchronization
 - 9.3.1 Introduction
 - 9.3.2 Frame Alignment Systems
 - 9.3.3 State Diagrams and Design Principles
 - 9.3.4 Choice of Frame Alignment Signal

DIGITAL CODING OF VOICE AND TELEVISION SIGNALS

- 10.1 Impairments Due to Sampling 10.1.1 Flat-top Samples 10.1.2 Anti-aliasing Filter 10.1.3 Reconstruction Filter 10.1.4 Switched Capacitor Filter Methods
- 10.2 Digital Encoding of Analogue Signals 10.2.1 Coding Methods 10.2.2 Distortion Criteria
- 10.3 PCM System Performance 10.3.1 Quantization Noise 10.3.2 Companding Techniques 10.3.3 CCITT Standards for Quantization Noise

10.4 Coding Techniques for Reduced Bit Rates 10.4.1 Principles of Data Compression 10.4.2 Adaptive Pulse Code Modulation (APCM) 10.4.3 Differential Pulse Code Modulation (DPCM) 10.4.4 Delta Modulation (DM) 10.4.5 Adaptive Differential Pu;se Code Modulation (ADPCM) 10.4.6 Adaptive Predictive Coding (APC)

ATTACHMENT 'A' (continued)

BASIC DIGITAL TRANSMISSION SYSTEM THEORY · COURSE OUTLINE

1. Linear Systems Dr K.C. Ng (Monash) 7 Lectures

Time domain and frequency domain, analysis of linear systems, including convolution and Fourier Transform. Application to Digital Transmission Systems.

2. Random Processes Dr K.K. Pang (Monash) 7 Lectures

Basic probability theory, random time processes including auto-correlation and power spectrum. Response of transmission systems to random inputs.

3. Digital Transmission Dr D.B. Keogh (Monash) 9 Lectures

Fundamentals.

Linear representation of digital signals. Bandlimiting, inter-symbol interference. Nyquist pulse shaping, roll-off. Average power spectral density of digital signals. Error probability formula for multilevel digital signals.

Line Coding.

Rationale. Examples - conditional diphase, AMI and HDB3. Physical description, average power spectral density.

Modulated digital signals.

Rationale. Examples, linear - digital Am (DSBAM, DSBAMSC, WAM, PM: VSBAM, SSBAM). Examples, nonlinear - digital FM. Time and frequency domain representations. Phase jitter tolerance. Application of baseband error probability formula to modulated digital signals.

4. <u>Applications</u> Dr A. Jennings and Messrs A. Martin & G. Nicholson (Telecom) 5 Lectures

Baseband transmission in pair cable. Effect of impulse noise and compatibility with other services on a 4800 bit/s DNN signal. 2048 kbit/s line systems - crosstalk into similar systems, noise figure.

Data-in-Voice (DIV) performance. 2048 kbit/s DIV transmission in 2 or 3 contiguous super-groups between Melbourne and Sydney.

Digital Radio. Example based on 140 Mbit/s digital radio system. Introduction single hop system design for flat fading, effect of frequency selective fading. Optical Fibre Systems Example based on 140 Mbit/s system. Noise processes, regenerator section length design.

5. Summing up and Review

General course revision and course evaluation.