

PRACTICAL SIGNAL PROCESSING

The principles of signal processing are fundamental to the operation of everyday devices such as digital cameras, mobile telephones and digital audio players. This book introduces the basic theory of digital signal processing, placing a strong emphasis on the use of techniques in real-world applications. The author uses intuitive arguments rather than mathematical ones wherever possible, reinforced by practical examples and diagrams.

The first part of the book covers sampling, quantisation, the Fourier transform, filters, Bayesian methods and numerical considerations. These ideas are then developed in the second part, illustrating how they are used in audio, image, and video processing and compression, and in communications. The book concludes with methods for the efficient implementation of algorithms in hardware and software. Throughout, links between various signal processing techniques are stressed and real-world examples showing the advantages and disadvantages of the different approaches are presented, enabling the reader to choose the best solution to a given problem.

With over 200 illustrations and over 130 exercises (including solutions), this book will appeal to practitioners working in any branch of signal processing, as well as to undergraduate students of electrical and computer engineering.

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Contents

Pı	reface	page ix	
Part I		Foundations	1
1	Intro	oduction	3
	1.1	What is a signal?	3
	1.2	Domain and range of a signal	3
	1.3	Converting signals from one form to another	4
	1.4	Processing signals	5
	1.5	Notation	6
2	Sampling		10
	2.1	Regular sampling	10
	2.2	What is lost in sampling?	11
	2.3	Examples of aliasing	13
	2.4	Negative frequencies	15
	2.5	The Nyquist limit	16
	2.6	Irregular sampling	17
3	Conversion between analogue and digital		25
	3.1	A simple digital signal processing system	25
	3.2	Non-linear quantisation	28
	3.3	How many bits do we need?	29
	3.4	Dither	30
	3.5	Non-ideal conversion	32
4	The frequency domain		37
	4.1	Measuring rotational speed	37
	4.2	More complicated motion	39
	4.3	Interpreting the Fourier transform	42

V



vi	Contents		
	Reconstructing a signal from its Fourier transform Real signals	43 43 44	
4.7 4.8	Dealing with non-periodic signals The fast Fourier transform	49 53	
	Filters 5.1 Smoothing a signal		
5.2 5.3	Analysing a filter Convolution in the frequency domain Correlation	63 66 74 76	
5.5	Designing FIR filters Interpolation Infinite impulse response filters	77 83 94 100	
6 Like 6.1 6.2	elihood methods	104 104 109 111	
7.1 7.2 7.3		124 124 135 139 143	
Part II	Applications	147	
8 Aud 8.1 8.2 8.3 8.4 8.5 8.6	The ear Sample rates and conversion Audio in the frequency domain Compression of audio signals	149 149 149 152 154 163	
9.1 9.2 9.3 9.4	Luminance and chrominance Gamma An image as a signal Filtering an image The discrete cosine transform	178 178 179 180 185	



		Contents	vii		
	9.6	JPEG compression for continuous-tone images	198		
	9.7	The discrete wavelet transform	202		
	9.8	Image scaling	209		
	9.9	Image enhancement	213		
	9.10	Edge detection	215		
	9.11	Processing of bilevel images	217		
	9.12	Pattern recognition	223		
10	Movin	231			
	10.1	Standard video formats	231		
	10.2	Deinterlacing	234		
	10.3	Standards conversion	239		
	10.4	Motion estimation	241		
	10.5	MPEG-2 video compression	246		
11	Communications		253		
	11.1	Amplitude modulation	254		
	11.2	Frequency modulation	260		
	11.3	Quadrature amplitude modulation	263		
	11.4	Spread spectrum schemes	287		
12	Impler	299			
	12.1	Software implementations	300		
	12.2	Processor architectures	302		
	12.3	Hardware implementations	307		
	12.4	Bit-serial arithmetic	313		
Ans	wers		323		
Ind	Index				



Preface

This is a book you can read in the park, on the beach, at the bus stop – or even in the bath.

The book is in two parts. The first part takes you step-by-step through the fundamental ideas of digital signal processing, while the second part shows how these ideas are used in a wide range of practical situations. My aim is that by the end of the book you will understand many of the signal processing algorithms and techniques that are essential to everyday devices such as digital cameras, modems, digital set-top boxes, mobile telephones and digital audio players. I have used examples drawn from the operation of such devices to help explain points in the text.

You do not need to know any calculus to understand any of the ideas discussed. A basic understanding of trigonometry and of arithmetic on complex numbers is necessary, however; and a very basic knowledge of the principles of electronic circuits is helpful, but by no means essential.

If you are a student, I hope that the approach this book takes will give you a more concrete and more intuitive grasp of the principles of digital signal processing than a purer mathematical treatment would. If you are a practising engineer or programmer with a particular problem to solve, I hope that the book helps you understand the problem and decide on the right way to tackle it. And if you are just interested in the subject for its own sake, I hope you enjoy the book.

There are exercises at the end of each chapter. Some of them you can probably do in your head; for some you might need pencil and paper; and for some you will need to write a short program. Some are slightly more ambitious programming projects. A few ask you to criticise inappropriate solutions to signal processing problems suggested by a hypothetical friend who has clearly not read this book: if you have any friends in this position you can remedy the situation by buying them a copy. Please try the easier exercises and at least think about how you would go about the harder ones; and please don't take your computer into the bath.



x Preface

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