

Signals and Communication Technology

For further volumes:
<http://www.springer.com/series/4748>

K.R. Rao • D.N. Kim •
J.J. Hwang

Fast Fourier Transform: Algorithms and Applications



Dr. K.R. Rao
Univ. of Texas at Arlington
Electr. Engineering
Nedderman Hall
Yates St. 416
76013 Arlington Texas
USA
rao@uta.edu

Dr. D.N. Kim
Univ. of Texas at Arlington
Electr. Engineering
Nedderman Hall
Yates St. 416
76013 Arlington Texas
USA
cooldnk@yahoo.com

Dr. J.J. Hwang
Kunsan National Univ.
School of Electron. & Inform.
Engineering
68 Miryong-dong
573-701 Kunsan
Korea, Republic of (South Korea)
hwang@kunsan.ac.kr

ISSN 1860-4862
ISBN 978-1-4020-6628-3 e-ISBN 978-1-4020-6629-0
DOI 10.1007/978-1-4020-6629-0
Springer Dordrecht Heidelberg London New York

Library of Congress Control Number: 2010934857

© Springer Science+Business Media B.V. 2010

No part of this work may be reproduced, stored in a retrieval system, or transmitted in any form or by any means, electronic, mechanical, photocopying, microfilming, recording or otherwise, without written permission from the Publisher, with the exception of any material supplied specifically for the purpose of being entered and executed on a computer system, for exclusive use by the purchaser of the work.

Cover design: SPi Publisher Services

Printed on acid-free paper

Springer is part of Springer Science+Business Media (www.springer.com)

Preface

This book presents an introduction to the principles of the fast Fourier transform (FFT). It covers FFTs, frequency domain filtering, and applications to video and audio signal processing.

As fields like communications, speech and image processing, and related areas are rapidly developing, the FFT as one of the essential parts in digital signal processing has been widely used. Thus there is a pressing need from instructors and students for a book dealing with the latest FFT topics.

This book provides a thorough and detailed explanation of important or up-to-date FFTs. It also has adopted modern approaches like MATLAB examples and projects for better understanding of diverse FFTs.

Fast Fourier transform (FFT) is an efficient implementation of the discrete Fourier transform (DFT). Of all the discrete transforms, DFT is most widely used in digital signal processing. The DFT maps a sequence either in the time domain or in the spatial domain into the frequency domain. The development of the DFT originally by Cooley and Tukey [A1] followed by various enhancements/modifications by other researchers has provided the incentive and the impetus for its rapid and widespread utilization in a number of diverse disciplines. Independent of the Cooley-Tukey approach, several algorithms such as prime factor, split radix, vector radix, split vector radix, Winograd Fourier transform, and integer FFT have been developed. The emphasis of this book is on various FFTs such as the decimation-in-time FFT, decimation-in-frequency FFT algorithms, integer FFT, prime factor DFT, etc.

In some applications such as dual-tone multi-frequency detection and certain pattern recognition, their spectra are skewed to some regions that are not uniformly distributed. With this basic concept we briefly introduce the nonuniform DFT (NDFT), dealing with arbitrarily spaced samples in the Z-plane, while the DFT deals with equally spaced samples on the unit circle with the center at the origin in the Z-plane.

A number of companies provide software for implementing FFT and related basic applications such as convolution/correlation, filtering, spectral analysis, etc. on various platforms. Also general-purpose DSP chips can be programmed to implement the FFT and other discrete transforms.

This book is designed for senior undergraduate and graduate students, faculty, engineers, and scientists in the field, and self-learners to understand FFTs and directly apply them to their fields, efficiently. It is designed to be both a text and a reference. Thus examples, projects and problems all tied with MATLAB, are provided for grasping the concepts concretely. It also includes references to books and review papers and lists of applications, hardware/software, and useful websites. By including many figures, tables, block diagrams and graphs, this book helps the reader understand the concepts of fast algorithms readily and intuitively. It provides new MATLAB functions and MATLAB source codes.

The material in this book is presented without assuming any prior knowledge of FFT. This book is for any professional who wants to have a basic understanding of the latest developments in and applications of FFT. It provides a good reference for any engineer planning to work in this field, either in basic implementation or in research and development.

D.N. Kim acknowledges the support by the National Information Technology (IT) Industry Promotion Agency (NIPA) and the Ministry of Knowledge Economy, Republic of Korea, under the IT Scholarship Program.

Organization of the Book

Chapter 1 introduces various applications of the discrete Fourier transform. Chapter 2 is devoted to introductory material on the properties of the DFT for the equally spaced samples. Chapter 3 presents fast algorithms to be mainly categorized as decimation-in-time (DIT) or decimation-in-frequency (DIF) approaches. Based on these, it introduces fast algorithms like split-radix, Winograd algorithm and others. Chapter 4 is devoted to integer FFT which approximates the discrete Fourier transform. One-dimensional DFT is extended to the two-dimensional signal and then to the multi-dimensional signal in Chapter 5. Applications to filtering are presented in this chapter. Variance distribution in the DFT domain is covered. It also introduces how we can diagonalize a circulant matrix using the DFT matrix. Fast algorithms for the 2-D DFT are covered in Chapter 6. Chapter 7 is devoted to introductory material on the properties of nonuniform DFT (NDFT) for the nonequally spaced samples. Numerous applications of the FFT are presented in Chapter 8. Appendix A covers performance comparison of discrete transforms. Appendix B covers spectral distance measures of image quality. Appendix C covers Integer DCTs. DCTs and DSTs are derived in Appendix D (DCT – discrete cosine transform, DST – discrete sine transform). Kronecker products and separability are briefly covered in Appendix E. Appendix F describes mathematical relations. Appendices G and H include MATLAB basics and M files. The bibliography contains lists of references to books and review papers, software/hardware, and websites. Numerous problems and projects are listed at the end of each chapter.

Contents

1	Introduction	1
1.1	Applications of Discrete Fourier Transform	2
2	Discrete Fourier Transform	5
2.1	Definitions	5
2.1.1	DFT	5
2.1.2	IDFT	5
2.1.3	Unitary DFT (Normalized)	6
2.2	The Z-Transform	7
2.3	Properties of the DFT	13
2.4	Convolution Theorem	18
2.4.1	Multiplication Theorem	24
2.5	Correlation Theorem	24
2.6	Overlap-Add and Overlap-Save Methods	27
2.6.1	The Overlap-Add Method	27
2.7	Zero Padding in the Data Domain	31
2.8	Computation of DFTs of Two Real Sequences Using One Complex FFT	32
2.9	A Circulant Matrix Is Diagonalized by the DFT Matrix	34
2.9.1	Toeplitz Matrix	34
2.9.2	Circulant Matrix	34
2.9.3	A Circulant Matrix Is Diagonalized by the DFT Matrix	35
2.10	Summary	37
2.11	Problems	37
2.12	Projects	40
3	Fast Algorithms	41
3.1	Radix-2 DIT-FFT Algorithm	42
3.1.1	Sparse Matrix Factors for the IFFT $N = 8$	46
3.2	Fast Algorithms by Sparse Matrix Factorization	47

3.3	Radix-2 DIF-FFT	56
3.3.1	DIF-FFT $N = 8$	57
3.3.2	In-Place Computations	61
3.4	Radix-3 DIT FFT	61
3.5	Radix-3 DIF-FFT	63
3.6	FFT for N a Composite Number	66
3.7	Radix-4 DIT-FFT	67
3.8	Radix-4 DIF-FFT	73
3.9	Split-Radix FFT Algorithm	75
3.10	Fast Fourier and BIFORE Transforms by Matrix Partitioning	78
3.10.1	Matrix Partitioning	78
3.10.2	DFT Algorithm	80
3.10.3	BT (BIFORE Transform)	82
3.10.4	CBT (Complex BIFORE Transform)	82
3.10.5	DFT (Sparse Matrix Factorization)	82
3.11	The Winograd Fourier Transform Algorithm	83
3.11.1	Five-Point DFT	83
3.11.2	Seven-Point DFT	84
3.11.3	Nine-Point DFT	85
3.11.4	DFT Algorithms for Real-Valued Input Data	85
3.11.5	Winograd Short- N DFT Modules	87
3.11.6	Prime Factor Map Indexing	88
3.11.7	Winograd Fourier Transform Algorithm (WFTA)	90
3.12	Sparse Factorization of the DFT Matrix	92
3.12.1	Sparse Factorization of the DFT Matrix Using Complex Rotations	92
3.12.2	Sparse Factorization of the DFT Matrix Using Unitary Matrices	94
3.13	Unified Discrete Fourier–Hartley Transform	97
3.13.1	Fast Structure for UDFHT	101
3.14	Bluestein’s FFT Algorithm	104
3.15	Rader Prime Algorithm	106
3.16	Summary	107
3.17	Problems	108
3.18	Projects	110
4	Integer Fast Fourier Transform	111
4.1	Introduction	111
4.2	The Lifting Scheme	112
4.3	Algorithms	112
4.3.1	Fixed-Point Arithmetic Implementation	117
4.4	Integer Discrete Fourier Transform	119
4.4.1	<i>Near-Complete</i> Integer DFT	119
4.4.2	<i>Complete</i> Integer DFT	121
4.4.3	Energy Conservation	123
4.4.4	Circular Shift	123

4.5 Summary	125
4.6 Problems	126
4.7 Projects	126
5 Two-Dimensional Discrete Fourier Transform	127
5.1 Definitions	127
5.2 Properties	131
5.2.1 Periodicity	131
5.2.2 Conjugate Symmetry	131
5.2.3 Circular Shift in Time/Spatial Domain (Periodic Shift)	133
5.2.4 Circular Shift in Frequency Domain (Periodic Shift)	133
5.2.5 Skew Property	135
5.2.6 Rotation Property	135
5.2.7 Parseval's Theorem	135
5.2.8 Convolution Theorem	136
5.2.9 Correlation Theorem	137
5.2.10 Spatial Domain Differentiation	139
5.2.11 Frequency Domain Differentiation	139
5.2.12 Laplacian	139
5.2.13 Rectangle	139
5.3 Two-Dimensional Filtering	140
5.3.1 Inverse Gaussian Filter (IGF)	142
5.3.2 Root Filter	142
5.3.3 Homomorphic Filtering	143
5.3.4 Range Compression	146
5.3.5 Gaussian Lowpass Filter	148
5.4 Inverse and Wiener Filtering	150
5.4.1 The Wiener Filter	151
5.4.2 Geometric Mean Filter (GMF)	154
5.5 Three-Dimensional DFT	156
5.5.1 3-D DFT	156
5.5.2 3-D IDFT	157
5.5.3 3D Coordinates	157
5.5.4 3-D DFT	157
5.5.5 3-D IDFT	157
5.6 Variance Distribution in the 1-D DFT Domain	158
5.7 Sum of Variances Under Unitary Transformation Is Invariant	160
5.8 Variance Distribution in the 2-D DFT Domain	160
5.9 Quantization of Transform Coefficients can be Based on Their Variances	162
5.10 Maximum Variance Zonal Sampling (MVZS)	166
5.11 Geometrical Zonal Sampling (GZS)	168
5.12 Summary	168
5.13 Problems	169
5.14 Projects	170

6	Vector-Radix 2D-FFT Algorithms	185
6.1	Vector Radix DIT-FFT	185
6.2	Vector Radix DIF-FFT	189
6.3	Summary	193
7	Nonuniform DFT	195
7.1	Introduction	195
7.2	One-Dimensional NDFT	196
7.2.1	DFT of Uniformly Sampled Sequences	196
7.2.2	Definition of the NDFT	197
7.2.3	Properties of the NDFT	200
7.2.4	Examples of the NDFT-2	203
7.3	Fast Computation of NDFT	208
7.3.1	Forward NDFT	208
7.3.2	Inverse NDFT	213
7.4	Two-Dimensional NDFT	217
7.4.1	2D Sampling Structure	218
7.4.2	Example of Two-Dimensional Nonuniform Rectangular Sampling	221
7.5	Filter Design Using NDFT	222
7.5.1	Low-Pass Filter Design	222
7.5.2	Example of Nonuniform Low-Pass Filter	230
7.6	Summary	233
7.7	Problems	233
8	Applications	235
8.1	Frequency Domain Downsampling	235
8.1.1	Frequency Domain Upsampling (Zero Insertion)	238
8.2	Fractal Image Compression	240
8.3	Phase Only Correlation	244
8.4	Image Rotation and Translation Using DFT/FFT	247
8.5	Intraframe Error Concealment	249
8.6	Surface Texture Analysis	250
8.7	FFT-Based Ear Model	251
8.8	Image Watermarking	251
8.9	Audio Watermarking	253
8.9.1	Audio Watermarking Using Perceptual Masking	255
8.10	OFDM	256
8.10.1	Signal Representation of OFDM Using IFFT/FFT	257
8.11	FFT Processors for OFDM	258
8.12	DF DFT-Based Channel Estimation Method	260
8.12.1	DF DFT-Based Channel Estimation Method	260
8.13	The Conjugate-Gradient Fast Fourier Transform (CG-FFT)	262
8.14	Modified Discrete Cosine Transform (MDCT)	262
8.15	Oddly Stacked TDAC	266

8.16	Perceptual Transform Audio Coder	273
8.17	OCF Coder	274
8.18	NMR Measurement System	274
8.19	Audio Coder for Mobile Reception	275
8.20	ASPEC (Adaptive Spectral Perceptual Entropy Coding of High Quality Music Signals)	277
8.21	RELP Vocoder (RELP: Residual Excited Linear Prediction) ...	278
8.22	Homomorphic Vocoders	278
8.23	MUSICAM (Masking-Pattern Universal Sub-Band Integrated Coding and Multiplexing)	280
8.24	AC-2 Audio Coder	280
8.25	IMDCT/IMDST Implementation via IFFT	283
8.26	MDCT/MDST Implementation via IFFT	287
8.27	Autocorrelation Function and Power Density Spectrum	287
8.27.1	Filtered White Noise	288
8.28	Three-Dimensional Face Recognition	289
8.29	Two-Dimesional Multirate Processing	291
8.29.1	Upsampling and Interpolation	292
8.29.2	Downsampling and Decimation	294
8.30	Fast Uniform Discrete Curvelet Transform (FUDCuT)	300
8.30.1	The Radon Transform	300
8.30.2	The Ridgelet Transform	301
8.30.3	The Curvelet Transform	303
8.31	Problems	311
8.32	Projects	314
8.32.1	Directional Bandpass Filters	315
Appendix A: Performance Comparison of Various Discrete Transforms		317
A.1	Transform Coding Gain	318
A.2	Variance Distribution in the Transform Domain	319
A.3	Normalized MSE	319
A.4	Rate Versus Distortion (Rate-Distortion)	319
A.5	Residual Correlation	320
A.6	Scalar Wiener Filtering	322
A.7	Geometrical Zonal Sampling (GZS)	323
A.8	Maximum Variance Zonal Sampling (MVZS)	324
Appendix B: Spectral Distance Measures of Image Quality		325
Project B		328
Appendix C: Integer Discrete Cosine Transform (INTDCT)		333
C.1	Integer DCT Via Lifting	333
C.1.1	Decomposition of DCT Via Walsh–Hadamard Transform	334
C.1.2	Implementation of Integer DCT	337

C.2 Integer DCT by the Principle of Dyadic Symmetry	337
C.2.1 Generation of the Order-Eight Integer DCT	338
C.2.2 Integer DCTs in Video Coding Standards	340
C.2.3 Performance of Eight-Point Integer DCTs	345
Problems	346
Projects	347
Appendix D: DCT and DST	349
D.1 Kernels for DCT and DST	349
D.2 Derivation of Unitary DCTs and DSTs	351
D.3 Circular Convolution Using DCTs and DSTs Instead of FFTs	359
D.4 Circular Shifting Property of the DCT	360
Problems	361
Projects	361
Appendix E: Kronecker Products and Separability	363
E.1 Kronecker Products	363
E.2 Generalized Kronecker Product	364
E.3 Separable Transformation	365
Appendix F: Mathematical Relations	367
Problem	368
Appendix G: Basics of MATLAB	369
G.1 List of MATLAB Related Websites	375
G.1.1 MATLAB Tutorial	375
G.1.2 MATLAB Commands and Functions	375
G.1.3 MATLAB Summary and Tutorial	375
G.1.4 A MATLAB Primer	375
G.1.5 MATLAB FAQ	375
G.2 List of MATLAB Related Books	375
Appendix H	379
H.1 MATLAB Code for 15-Point WFTA	379
H.2 MATLAB Code for Phase Only Correlation (POC)	382
Bibliography	383
Index	415

Abbreviations

AAC	Advanced audio coder, AAC-LD (low delay) AAC-LC (low complexity)
AC, AC-2, AC-3	Audio coder
ACATS	FCC advisory committee on advanced television service
ACM	Association for computing machinery
ADSL	Asymmetric digital subscriber line
AES	Audio engineering society
ANN	Artificial neural network
ANSI	American National Standards Institute
APCCAS	IEEE Asia-Pacific Conference on Circuits and Systems
ASIC	Application specific integrated circuit
ASPEC	Adaptive spectral perceptual entropy coding of high quality music signals
ASSP	Acoustics, speech, and signal processing
ATC	Adaptive transform coding
ATRAC	Adaptive transform acoustic coding
ATSC	Advanced television systems committee
AVC	Advanced video coding, MPEG-4 AVC (MPEG-4 Part 10)
AVS	Audio video standard
BER	Bit error rate
BF	Butterfly
BIFORE	Binary Fourier representation
BPF	Band pass filter
BRO	Bit reversed order
BV	Basis vector
CAS	Circuits and systems
CBT	Complex BIFORE transform
CCETT	<i>Centre Commun d'Etudes de Télédiffusion et Télécommunications</i> in French (Common Study Center of Telediffusion and Telecommunications)

CD	Compact disc
CDMA	Code division multiple access
CE	Consumer electronics
CELP	Code-excited linear prediction
CF	Continuous flow
CFA	Common factor algorithms
CGFFT	Conjugate gradient FFT
CICC	Custom integrated circuits conference
CIPR	Center for image processing research
CMFB	Cosine modulated filter bank
CODEC	Coder and decoder
COFDM	Coded orthogonal frequency division multiplex
CM	Circulant matrix
CR	Compression ratio
CRT	Chinese remainder theorem
CSVT	Circuits and systems for video technology
DA	Distributed arithmetic
DAB	Digital audio broadcasting
DCC	Digital compact cassette
DCT	Discrete cosine transform
DEMUX	Demultiplexer
DF	Decision feedback
DFT	Discrete Fourier transform
DHT	Discrete Hartley transform
DIS	Draft International Standard
DMT	Discrete multitone modulation
DOS	Disc operating systems
DOT	Discrete orthogonal transform
DPCM	Differential pulse code modulation
D-PTS	Decomposition partial transmit sequence
DSP	Digital signal processing/processor
DST	Discrete sine transform
DTMF	Dual-tone multifrequency. DTMF signals/tones are used for telephone touch keypads [B40]
DTT	Discrete trigonometric transform
DVB-T	Digital video broadcasting standard for terrestrial transmission, using COFDM modulation
DVD	Digital video/versatile disc
DWT	Discrete wavelet transform
ECCTD	Biennial European Conference on Circuit Theory and Design
EDN	Electrical design news
EECON	Electrical Engineering Conference of Thailand
EEG	Electroencephalograph
EKG	Electrocardiograph, ECG

EMC	Electromagnetic compatibility, IEEE transactions on
EURASIP	European Association for Signal Processing
EUSIPCO	European Signal Processing Conference
FAQ	Frequently asked questions
FCC	The Federal Communications Commission
FFT	Fast Fourier transform
FIR	Finite impulse response
FMM	Fast multipole method
FPGA	Field programmable gate array
FRAT	Finite radon transform
FRExt	Fidelity range extensions
FRIT	Finite ridgelet transform
FSK	Frequency shift keying
FTP	File transfer protocol
FUDCuT	Fast uniform discrete curvelet transform
FxpFFT	Fixed point FFT
GDFHT	Generalized discrete Fourier Hartley transform
GDFT	Generalized DFT
GLOBECOM	IEEE Global Telecommunications Conference
GZS	Geometrical zonal sampling
H.263	Standard for visual communication via telephone lines
HDTV	High-definition television
HPF	High pass filter
HT	Hadamard transform
HTTP	Hyper text transfer protocol
HVS	Human visual sensitivity
Hz	Hertz cycles/sec, or cycles/meter
Iasted	The International Association of Science and Technology for Development
IBM	International business machines
IC	Integrated circuit(s)
ICA	Independent component analysis
ICASSP	IEEE International Conference on Acoustics, Speech, and Signal Processing
ICC	IEEE International Conference on Communications
ICCE	IEEE International Conference on Consumer Electronics
ICCS	IEEE International Conference on Circuits and Systems
ICECS	International Conference on Electronics, Circuits and Systems
ICIP	IEEE International Conference on Image Processing
ICME	IEEE International Conference on Multimedia and Expo
ICSPAT	International Conference on Signal Processing Applications and Technology
IDFT	Inverse DFT
IEC	International Electrotechnical Commission

IEEE	The Institute of Electrical and Electronics Engineers
IEICE	Institute of Electronics, Information and Communication Engineers
IFFT	Inverse FFT
IGF	Inverse Gaussian filter
IJG	Independent JPEG group
ILPM	Inverse LPM (log polar mapping)
IMTC	IEEE Instrumentation and Measurement Technology Conference
IntFFT	Integer FFT
IP	Image processing / intellectual property
IRE	Institute of radio engineers
IS	International standard
IS&T	The Society for Imaging Science and Technology
ISCAS	IEEE International Symposium on Circuits and Systems
ISCIT	IEEE International Symposium on Communications and Information Technologies
ISDN	Integrated services digital network
ISO	International Organization for Standardization
ISPACS	IEEE International Symposium on Intelligent Signal Processing and Communication Systems
IT	Information theory
JPEG	Joint photographic experts group
JSAC	Journal on selected areas in communications, IEEE
JTC	Joint technical committee
KLT	Karhunen–Loëve transform
LAN	Local area networks
LC	Low complexity
LMS	Least mean square
LO	Lexicographic ordering
LPF	Low pass filter
LPM	Log polar mapping
LPOS	Left point of symmetry
LS	Least square, lifting scheme
LSI	Linear shift invariant
LUT	Look up tables
LW	Long window
MCM	Multichannel carrier modulation
MD	MiniDisc
MDCT	Modified discrete cosine transform
MDST	Modified discrete sine transform
ML	Maximum likelihood/multiplierless
MLS	Maximum length sequence
MLT	Modulated lapped transform
MMSE	Minimum mean square error

MoM	The method of moments
MOPS	Million operations per second
MOS	Mean opinion score
MOV	Model output variable
MPEG	Moving picture experts group
MR	Mixed radix
MRI	Magnetic resonance imaging
ms	Millisecond
M/S	Mid/side, middle and side, or sum and difference
MSB	Most significant bit
MUSICAM	Masking pattern universal subband integrated coding and multiplexing (MPEG-1 Level 2, MP2)
MUX	Multiplexer
MVP	Multimedia video processor
MWSCAS	Midwest Symposium on Circuits and Systems
NAB	National Association of Broadcasters
NBC	Nonbackward compatible (with MPEG-1 audio)
NMR	Noise-to-mask ratio
NNMF	Nonnegative matrix factorization
NTC	National telecommunications conference
OCF	Optimum coding in the frequency domain
OEM	Original equipment manufacturer
OFDM	Orthogonal frequency-division multiplexing
ONB	Orthonormal basis
PAC	Perceptual audio coder
PAMI	Pattern analysis and machine intelligence
PAPR	Peak-to-average power ratio
PC	Personal computer
PCM	Pulse code modulation
PDPTA	International Conference on Parallel and Distributed Processing Techniques and Applications
PE	Perceptual entropy
PFA	Prime factor algorithm
PFM	Prime factor map
PoS	Point of symmetry
PQF	Polyphase quadrature filter
PR	Perfect reconstruction
PRNG	Pseudorandom number generator
P/S	Parallel to serial converter
PSF	Point spread function
PSK	Phase shift keying
PSNR	Peak-to-peak signal-to-noise ratio
QAM	Quadrature amplitude modulation
QMF	Quadrature mirror filter

QPSK	Quadrature phase-shift keying
RA	Radon transform
RAM	Random access memory
RELP	Residual excited linear prediction
RF	Radio frequency
RFFT	Real valued FFT
RI	Ridgelet transform
RMA	Royal Military Academy of Belgium
RPI	Rensselaer Polytechnic Institute
RPOS	Right point of symmetry
R–S	Reed Solomon
RST	Rotation, scaling and translation
SDDS	Sony dynamic digital sound
SEPXFM	Stereo-entropy-coded perceptual transform coder
SIAM	Society for Industrial and Applied Mathematics
SiPS	Signal processing systems
SMPTE	Society of Motion Picture and Television Engineers
SMR	Signal-to-mask ratio
SNR	Signal to noise ratio
SOPOT	Sum-of-powers-of-two
SP	Signal processing
S/P	Serial to parallel converter
SPIE	Society of Photooptical and Instrumentation Engineers
SPS	Symmetric periodic sequence
SR	Split radix
SS	Spread spectrum
SSST	Southeastern Symposium on System Theory
STBC	Space-time block code
SW	Short window
TDAC	Time domain aliasing cancellation
T/F	Time-to-frequency
UDFHT	Unified discrete Fourier-Hartley transform
USC	University of Southern California
VCIP	SPIE and IS&T visual communications and image processing
VCIR	Visual communication and image representation
VLSI	Very large scale integration
VSP	Vector signal processor
WD	Working draft
WHT	Walsh-Hadamard transform
WLAN	Wireless LAN
WMV	Window media video
WPMC	International Symposium on Wireless Personal Multimedia Communications