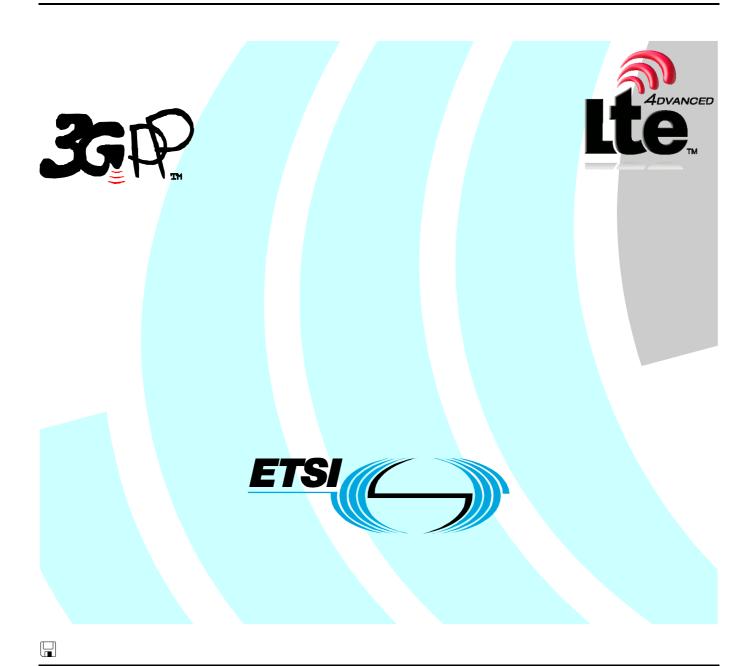
ETSI TS 126 410 V10.0.0 (2011-04)

Technical Specification

Digital cellular telecommunications system (Phase 2+);
Universal Mobile Telecommunications System (UMTS);
LTE;
General audio codec audio processing functions;
Enhanced aacPlus general audio codec;

Floating-point ANSI-C code (3GPP TS 26.410 version 10.0.0 Release 10)



Reference RTS/TSGS-0426410va00 Keywords GSM, LTE, UMTS

ETSI

650 Route des Lucioles F-06921 Sophia Antipolis Cedex - FRANCE

Tel.: +33 4 92 94 42 00 Fax: +33 4 93 65 47 16

Siret N° 348 623 562 00017 - NAF 742 C Association à but non lucratif enregistrée à la Sous-Préfecture de Grasse (06) N° 7803/88

Important notice

Individual copies of the present document can be downloaded from: <u>http://www.etsi.org</u>

The present document may be made available in more than one electronic version or in print. In any case of existing or perceived difference in contents between such versions, the reference version is the Portable Document Format (PDF). In case of dispute, the reference shall be the printing on ETSI printers of the PDF version kept on a specific network drive within ETSI Secretariat.

Users of the present document should be aware that the document may be subject to revision or change of status.

Information on the current status of this and other ETSI documents is available at

http://portal.etsi.org/tb/status/status.asp

If you find errors in the present document, please send your comment to one of the following services: http://portal.etsi.org/chaircor/ETSI_support.asp

Copyright Notification

No part may be reproduced except as authorized by written permission. The copyright and the foregoing restriction extend to reproduction in all media.

© European Telecommunications Standards Institute 2011. All rights reserved.

DECTTM, **PLUGTESTS**TM, **UMTS**TM, **TIPHON**TM, the TIPHON logo and the ETSI logo are Trade Marks of ETSI registered for the benefit of its Members.

3GPP[™] is a Trade Mark of ETSI registered for the benefit of its Members and of the 3GPP Organizational Partners. **LTE**[™] is a Trade Mark of ETSI currently being registered

for the benefit of its Members and of the 3GPP Organizational Partners.

GSM® and the GSM logo are Trade Marks registered and owned by the GSM Association.

Intellectual Property Rights

IPRs essential or potentially essential to the present document may have been declared to ETSI. The information pertaining to these essential IPRs, if any, is publicly available for **ETSI members and non-members**, and can be found in ETSI SR 000 314: "Intellectual Property Rights (IPRs); Essential, or potentially Essential, IPRs notified to ETSI in respect of ETSI standards", which is available from the ETSI Secretariat. Latest updates are available on the ETSI Web server (http://webapp.etsi.org/IPR/home.asp).

Pursuant to the ETSI IPR Policy, no investigation, including IPR searches, has been carried out by ETSI. No guarantee can be given as to the existence of other IPRs not referenced in ETSI SR 000 314 (or the updates on the ETSI Web server) which are, or may be, or may become, essential to the present document.

Foreword

This Technical Specification (TS) has been produced by ETSI 3rd Generation Partnership Project (3GPP).

The present document may refer to technical specifications or reports using their 3GPP identities, UMTS identities or GSM identities. These should be interpreted as being references to the corresponding ETSI deliverables.

The cross reference between GSM, UMTS, 3GPP and ETSI identities can be found under http://webapp.etsi.org/key/queryform.asp.

Contents

History		24
Annex B	informative): Change history	22
Annex A	informative): Weighted MOPS and PROM	21
5.3	Error pattern file (decoder input)	20
	Bitstream file format (encoder output/decoder input)	
	Audio input file (encoder input/decoder output)	
	formats	
4.3.4	Maximum stack size	16
4.3.3	Dynamic memory	
4.3.2	Static memory	
4.3.1	Constants and tables	9
	Memory requirements	
	Program execution	
	Contents of the floating point ANSI-C source code	
4 Flo	ting point ANSI-C code structure	6
3.2	Abbreviations	5
3.1	Definitions	5
3 De	nitions and abbreviations	5
2 Re	erences	5
1 Sco	pe	5
Foreword		2
Intellectu	l Property Rights	2

Foreword

This Technical Specification has been produced by the 3rd Generation Partnership Project (3GPP).

The contents of the present document are subject to continuing work within the TSG and may change following formal TSG approval. Should the TSG modify the contents of the present document, it will be re-released by the TSG with an identifying change of release date and an increase in version number as follows:

Version x.y.z

where:

- x the first digit:
 - 1 presented to TSG for information;
 - 2 presented to TSG for approval;
 - 3 or greater indicates TSG approved document under change control.
- y the second digit is incremented for all changes of substance, i.e. technical enhancements, corrections, updates, etc.
- z the third digit is incremented when editorial only changes have been incorporated in the document.

1 Scope

The present document contains an electronic copy of the ANSI-C code for the Floating-point Enhanced aacPlus codec [1].

2 References

The following documents contain provisions which, through reference in this text, constitute provisions of the present document.

- References are either specific (identified by date of publication, edition number, version number, etc.) or non-specific.
- For a specific reference, subsequent revisions do not apply.
- For a non-specific reference, the latest version applies. In the case of a reference to a 3GPP document (including a GSM document), a non-specific reference implicitly refers to the latest version of that document *in the same Release as the present document*.
- [1] 3GPP TS 26.401: "Enhanced aacPlus general audio codec; General Description". [2] 3GPP TS 26.403: "Enhanced aacPlus general audio codec; Encoder Specification AAC part". 3GPP TS 26.404: "Enhanced aacPlus general audio codec; Encoder Specification SBR part". [3] [4] 3GPP TS 26.405: "Enhanced aacPlus general audio codec; Encoder Specification Parametric Stereo part". [5] ISO/IEC 14496-3:2001: "Information technology - Coding of audio-visual objects - Part 3: Audio". [6] ISO/IEC 14496-3:2001/Amd.1:2003: "Bandwidth Extension". [7] ISO/IEC 14496-3:2001/Amd.1:2003/DCOR1". [8] ISO/IEC 14496-3:2001/ Amd.2:2004: "Parametric Coding for High Quality Audio. [9] 3GPP TS 26.402: Enhanced aacPlus general audio codec; Additional Decoder Tools".

3 Definitions and abbreviations

3.1 Definitions

For the purposes of the present document, the terms and definitions given in TS 26.401 [1], TS 26.403 [2], TS 26.404 [3], TS 26.405 [4] and TS 26.402 [9] apply.

3.2 Abbreviations

For the purposes of the present document, the following abbreviations apply:

AAC Advanced Audio Coding

aacPlus Combination of MPEG-4 AAC and MPEG-4 Bandwidth extension (SBR)

Enhanced aacPlus Combination of MPEG-4 AAC, MPEG-4 Bandwidth extension (SBR) and MPEG-4

Parametric Stereo

MDCT Modified Discrete Cosine Transform

QMF Quadrature Mirror Filter SBR Spectral Band Replication

ANSI	American National Standards Institute
GSM	Global System for Mobile communications

I/O Input/Output

RAM Random Access Memory ROM Read Only Memory

4 Floating point ANSI-C code structure

This clause gives an overview of the structure of the floating point ANSI-C code and provides an overview of the contents and organization of the C code attached to the present document.

The C code has been verified on the following systems:

- IBM PC/AT compatible computers with Windows XP, 2000 and Microsoft Visual C++ v.6.0 compiler.
- IBM PC/AT compatible computers with Linux OS and GCC v.3.3 compiler.

ANSI-C was selected as the programming language because portability was desirable.

4.1 Contents of the floating point ANSI-C source code

The C code distribution is organised in two directories for encoder and decoder and further into several subdirectories, reflecting the major building blocks of the Enhanced aacPlus codec. The file descriptions on root level as well as the directory structure is given as follows:

Table 1: Source code directory structure for the encoder (FloatFR_aacPlusenc)

Directory	Description
README.txt	information on how to compile
Makefile	UNIX style encoder Makefile
FloatFR_aacPlusEnc.dsw	Win32 MSVC 6.0 encoder workspace
FloatFR_aacPlusEnc.dsp	Win32 MSVC 6.0 encoder makefile
src/	directory for the encoder frontend
	·
FloatFR_fastaacenc/	AAC encoder library
FloatFR_resamplib/	resampler library
FloatFR_sbrenclib/	SBR encoder library

Table 2: Source code directory structure for the decoder (FloatFR_aacPlusdec)

Directory	Description
README.txt	information on how to compile
Makefile	UNIX style encoder Makefile
FloatFR_aacPlusdec_mp eg4.dsw	Win32 MSVC 6.0 decoder workspace
FloatFR_aacPlusdec_mp eg4.dsp	Win32 MSVC 6.0 decodec makefile
src/	directory for the decoder frontend
FloatFR_aacdec	AAC decoder library
FloatFR_sbrdeclib/	SBR decoder library

Table 3: Source code directory structure common for encoder and decoder

Directory	Description			
FloatFR_bitbuflib/	bitstream reading/writing library			
FloatFRlib/	general purpose functionalities			
lib/	precompiled libraries for audio and bitstream			
	file format handling			

The distributed files with suffix "c" contain the source code and the files with suffix "h" are the header files. Within the respective libraries, the RAM data is contained in "xxx_ram" files with suffix "c", the ROM data is contained in "xxx_rom" files with suffix "c". Makefiles are provided for the platforms in which the C code has been verified (listed above).

Note that the FloatFRlib/, FloatFR_bitbuflib/ and lib/ directory are identical for encoder and decoder. A list of source code files with the respective lines of code (pure C instructions) is given below:

Table 4: Encoder source code files and lines of code

Directory	Module	Lines of code		
•				
src/	main.c	332		
EL (ED () III /	mp4file.c	255		
FloatFR_fastaacenclib/	qc_main.c	224		
	aacenc.c	136		
	ms_stereo.c	50		
	spreading.c	10		
	interface.c	44		
	bit_cnt.c	588		
	adj_thr.c	592		
	quantize.c	56		
	psy_configuration.c	175		
	sf_estim.c	508		
	tns_param.c	45		
	grp_data.c	114		
	pre_echo_control.c	22		
	stprepro.c	149		
	tns.c	358		
	dyn_bits.c	281		
	psy_main.c	232		
	channel_map.c	52		
	block_switch.c	201		
	band_nrg.c	34		
	transform.c	151		
	bitenc.c	262		
	line_pe.c	55		
	stat_bits.c	107		
FloatFR_sbrenclib/	qmf_enc.c	565		
	ton_corr.c	287		
	fram_gen.c	688		
	env_bit.c	56		
	env_est.c	630		
	mh_det.c	515		
	hybrid.c	139		
	bit_sbr.c	375		
	ps_bitenc.c	225		
	sbr_main.c	355		
	tran_det.c	183		
	sbr_misc.c	49		
	code_env.c	290		
	nf_est.c	195		
	freq_sca.c	309		
	_	299		
FloatFR resamplib/		_		
		68		
FloatFR_resamplib/	invf_est.c ps_enc.c iir32resample.c resampler.c	140 299 71		

Table 5: Decoder source code files and lines of code

Directory	Module	Lines of code
src/	main.c	299
	fileifc.c	173
	spline_resampler.c	172
FloatFR_aacdec/	aacdecoder.c	172
	streaminfo.c	10
	channelinfo.c	102
	stereo.c	78
	longblock.c	234
	shortblock.c	241
	pulsedata.c	24
	block.c	163
	pns.c	89
	imdct.c	50
	tns.c	137
	bitstream.c	15
	channel.c	92
	conceal.c	245
	dse.c	9
FloatFR_sbrdeclib/	env_dec.c	370
	FFR_aacPLUScheck.c	32
	sbr_bitb.c	37
	env_calc.c	775
	lpp_tran.c	504
	sbrdecoder.c	514
	sbr_dec.c	218
	sbr_crc.c	45
	sbr_fft.c	615
	hybrid.c	140
	ps_bitdec.c	223
	huff_dec.c	9
	env_extr.c	655
	freq_sca.c	337
	ps_dec.c	317
	qmf_dec.c	526

Table 6: Common source code files and lines of code

Directory	Module	Lines of code
FloatFR_bitbuflib/	bitbuffer.c	111
FloatFRlib/	cfftn.c	649
	transcendent.c	15

4.2 Program execution

The Enhanced aacPlus codec is implemented in two programs:

- enhAacPlusEnc.exe
- enhAacPlusDec.exe

The programs should be called like:

- enhAacPlusEnc.exe <wav_file> <bitstream_file> <bitrate> <(m)ono/(s)tereo>
- enhAacPlusDec.exe

bitstream_file> <wav_file> <mode> [error_pattern_file]

The audio files contain 16-bit linear encoded PCM samples with wav header, the bitstream files are of 3GPP type an the error patter file is a ASCII file, see section 5.

The encoder and decoder command line handling is also explained by running the applications without input arguments.

4.3 Memory requirements

The data types of variables and tables used in the floating-point implementation are plain ANSI-C data types, the following types are used:

- char
- unsigned char
- short
- int
- unsigned int
- float

4.3.1 Constants and tables

This clause contains a listing of all constants and tables contributing to the ROM requirements of the encoder and decoder.

Table 7: Encoder constants and tables

Name	Data	Size	Allocated in	Description
	type	[word]	Source File	2000. P 11011
LongWindowSine	float	1024	aac_rom.c	Window coefficients
ShortWindowSine	float	128	aac_rom.c	Window coefficients
LongWindowKBD	float	1024	aac_rom.c	Window coefficients
fftTwiddleTab	float	513	aac_rom.c	FFT twiddle coefficients
quantTableQ	float	16	aac_rom.c	Quantizer table, used for efficient pow ()
				implementation
quantTableE	float	17	aac_rom.c	Quantizer table, used for efficient pow ()
		40		implementation
invQuantTableQ	float	16	aac_rom.c	Quantizer table, used for efficient pow ()
invQuantTableE	float	17	aac_rom.c	implementation Quantizer table, used for efficient pow ()
IIIVQualitTableL	iioat	17	aac_rom.c	implementation
pow4_3_tab	float	64	aac_rom.c	Quantizer table, used for efficient pow ()
	noat		ddo_rom.o	implementation
p_8000_mono_long	float	4	aac_rom.c	TNS tuning parameters
p_8000_stereo_long	float	4	aac_rom.c	TNS tuning parameters
p_8000_mono_short	float	4	aac_rom.c	TNS tuning parameters
p_8000_stereo_short	float	4	aac_rom.c	TNS tuning parameters
p_16000_mono_long	float	4	aac_rom.c	TNS tuning parameters
p_16000_stereo_long	float	4	aac_rom.c	TNS tuning parameters
p_16000_mono_short	float	4	aac_rom.c	TNS tuning parameters
p_16000_stereo_short	float	4	aac_rom.c	TNS tuning parameters
p_24000_mono_long	float	4	aac_rom.c	TNS tuning parameters
p_24000_stereo_long	float	4	aac_rom.c	TNS tuning parameters
p_24000_mono_short	float	4	aac_rom.c	TNS tuning parameters
p_24000_stereo_short	float	4	aac_rom.c	TNS tuning parameters
p_32000_mono_long	float	4	aac_rom.c	TNS tuning parameters
p_32000_stereo_long	float float	4	aac_rom.c	TNS tuning parameters
p_32000_mono_short p_32000_stereo_short	float	4	aac_rom.c	TNS tuning parameters TNS tuning parameters
tnsCoeff3	float	8	aac_rom.c aac_rom.c	TNS filter coefficients
tnsCoeff3Borders	float	8	aac_rom.c	TNS filter borders
tnsCoeff4	float	16	aac_rom.c	TNS filter coefficients
tnsCoeff4Borders	float	16	aac_rom.c	TNS filter borders
tnsInfoTab	int	24	aac_rom.c	TNS bitrate to tuning mapping table
tnsMaxBandsTab	int	27	aac_rom.c	max. TNS bands per sampling rate table
huff_ltab1_2	short	80	aac_rom.c	Huffman codeword table AAC
huff_ltab3_4	short	80	aac_rom.c	Huffman codeword table AAC
huff_ltab5_6	short	80	aac_rom.c	Huffman codeword table AAC
huff_ltab7_8	short	64	aac_rom.c	Huffman codeword table AAC
huff_ltab9_10	short	168	aac_rom.c	Huffman codeword table AAC
huff_ltab11	short	288	aac_rom.c	Huffman codeword table AAC
huff_ltabscf	short	120	aac_rom.c	Huffman codeword table AAC
huff_ctab1	short	80	aac_rom.c	Huffman codeword table AAC
huff_ctab2	short	80	aac_rom.c	Huffman codeword table AAC
huff_ctab3	short	80	aac_rom.c	Huffman codeword table AAC
huff_ctab4	short	80	aac_rom.c	Huffman codeword table AAC
huff_ctab5 huff_ctab6	short	80	aac_rom.c	Huffman codeword table AAC
huff_ctab7	short short	80 64	aac_rom.c aac_rom.c	Huffman codeword table AAC Huffman codeword table AAC
huff_ctab8	short	64	aac_rom.c	Huffman codeword table AAC
huff_ctab9	short	168	aac_rom.c	Huffman codeword table AAC
huff_ctab10	short	168	aac_rom.c	Huffman codeword table AAC
huff_ctab11	short	288	aac_rom.c	Huffman codeword table AAC
huff_ctabscf	short	242	aac_rom.c	Huffman codeword table AAC
sfb_11025_long_1024	char	43	aac_rom.c	Scalefactor band table
sfb_11025_short_128	char	15	aac_rom.c	Scalefactor band table
sfb_12000_long_1024	char	43	aac_rom.c	Scalefactor band table
sfb_12000_short_128	char	15	aac_rom.c	Scalefactor band table
sfb_16000_long_1024	char	43	aac_rom.c	Scalefactor band table
sfb_16000_short_128	char	15	aac_rom.c	Scalefactor band table
sfb_22050_long_1024	char	47	aac_rom.c	Scalefactor band table

	able			
J	able	_•.		
tuningTable	tuningT	231	sbr_main.c	SBR tuning parameters
coeffDen	float	8	iir32resample.c	
coeffNum	float	8	iir32resample.c	IIR filter coefficients for 3:2 resampling
set1_b	float	5	resampler.c	IIR filter coefficients for 2:1 resampling
set1_a set1_b	float	14	resampler.c	IIR filter coefficients for 2:1 resampling
logDualisTable	float float	65 14	transcendent.c resampler.c	Lookup table for efficient log() implementation IIR filter coefficients for 2:1 resampling
aBookPsIccTimeCode;	short	15	sbr_rom.c	Huffman codeword table Parametric Stereo
aBookPslccFreqCode	short	15	sbr_rom.c	Huffman codeword table Parametric Stereo
bookSbrEnvBalanceL11F	char	25	sbr_rom.c	Huffman codeword table SBR
v_Huff_envelopeLevelL11F	char	63	sbr_rom.c	Huffman codeword table SBR
bookSbrNoiseBalanceL11T	char	25	sbr_rom.c	Huffman codeword table SBR
v_Huff_NoiseLevelL11T	char	63	sbr_rom.c	Huffman codeword table SBR
bookSbrEnvBalanceL11T	char	25	sbr_rom.c	Huffman codeword table SBR
v_Huff_envelopeLevelL11T	char	63	sbr_rom.c	Huffman codeword table SBR
bookSbrEnvBalanceL10T	char	49	sbr_rom.c	Huffman codeword table SBR
bookSbrEnvBalanceL10F	char	49	sbr_rom.c	Huffman codeword table SBR
v_Huff_envelopeLevelL10F	char	121	sbr_rom.c	Huffman codeword table SBR
v_Huff_envelopeLevelL10T	char	121	sbr_rom.c	Huffman codeword table SBR
aBookPsIccTimeLength	char	15	sbr_rom.c	Huffman codeword table Parametric Stereo
aBookPslccFreqLength	char	15	sbr_rom.c	Huffman codeword table Parametric Stereo
aBookPslidFreqLength	char	29	sbr_rom.c	Huffman codeword table Parametric Stereo
aBookPslidTimeLength	char	29	sbr_rom.c	Huffman codeword table Parametric Stereo
bookSbrEnvBalanceC11F	int	25	sbr_rom.c	Huffman codeword table SBR
v_Huff_envelopeLevelC11F	int	63	sbr_rom.c	Huffman codeword table SBR
bookSbrNoiseBalanceC11T	int	25	sbr_rom.c	Huffman codeword table SBR
bookSbrEnvBalanceC11T	int	25	sbr_rom.c	Huffman codeword table SBR
v_Huff_NoiseLevelC11T	int	63	sbr_rom.c	Huffman codeword table SBR
bookSbrEnvBalanceC10T v_Huff_envelopeLevelC11T	int int	49 63	sbr_rom.c sbr_rom.c	Huffman codeword table SBR Huffman codeword table SBR
bookSbrEnvBalanceC10F	int	49	sbr_rom.c	Huffman codeword table SBR
v_Huff_envelopeLevelC10F	int	121	sbr_rom.c	Huffman codeword table SBR
v_Huff_envelopeLevelC10T	int	121	sbr_rom.c	Huffman codeword table SBR
11 " 1 1040T		404		Stereo groups
bins2groupMap	int	29	sbr_rom.c	Mapping of Parametric Stereo bins to Parametric
groupBordersMix	int	29	sbr_rom.c	Borders of Parametric Stereo groups
hiResBandBorders	int	21	sbr_rom.c	Borders of Parametric Stereo bins
aHybridResolution	int	3	sbr_rom.c	Number of hybrid bands in each QMF band
aBookPslidFreqCode	int	29	sbr_rom.c	Huffman codeword table Parametric Stereo
aBookPsIidTimeCode	int	29	sbr_rom.c	Huffman codeword table Parametric Stereo
trigData_fct4_8	float	8	sbr_rom.c	FFT twiddle table
trigData_fct4_16	float	16	sbr_rom.c	FFT twiddle table
trigData_fct4_32	float	32	sbr_rom.c	FFT twiddle table
b_0 :_0 :0_q		0.0	oo	obsolete)
p_64_640_qmf	float	640	sbr_rom.c	QMF window coefficients (Note: could be made
sbr_qmf_64_640	float	325	sbr_rom.c	QMF window coefficients
sbr_alt_sin_twiddle	float	17	sbr_rom.c	QMF filterbank twiddle table
sbr_cos_twiddle sbr_sin_twiddle	float	16	sbr_rom.c sbr_rom.c	QMF filterbank twiddle table QMF filterbank twiddle table
p8_13	float float	13 16	sbr_rom.c	Hybrid filterbank coefficients
p4_13	float	13	sbr_rom.c	Hybrid filterbank coefficients
saClass	float	7	sbr_rom.c	Parametric Stereo quantization table
panClass	float	7	sbr_rom.c	Parametric Stereo quantization table
sfb_24000_short_128	char	15	aac_rom.c	Scalefactor band table
sfb_24000_long_1024	char	47	aac_rom.c	Scalefactor band table
-fl- 04000 I 4004				

Table 8: Decoder constants and tables

Name	Data	Size	Allocated in	Description
	type	[word]	Source File	
tnsCoeff3	float	8	aac_rom.c	TNS filter coefficients

tnoCooff4	£1 = - 4	40	000 ========	TNC filter coefficients
tnsCoeff4	float	16 513	aac_rom.c	TNS filter coefficients
trigData	float float	1024	aac_rom.c	Sine table, used for efficient sin(), cos() Window coefficients
OnlyLongWindowKBD OnlyShortWindowKBD	float	1024	aac_rom.c	Window coefficients Window coefficients
,	_	1024	aac_rom.c	
OnlyLongWindowSine OnlyShortWindowSine	float		aac_rom.c	Window coefficients
	float	128	aac_rom.c	Window coefficients
sfb_48_1024	short	50	aac_rom.c	Scalefactor band table
sfb_48_128	short	15	aac_rom.c	Scalefactor band table
sfb_32_1024	short	51	aac_rom.c	Scalefactor band table
sfb_24_1024	short	49	aac_rom.c	Scalefactor band table
sfb_24_128	short	16	aac_rom.c	Scalefactor band table
sfb_16_1024	short	44	aac_rom.c	Scalefactor band table
sfb_16_128	short	16	aac_rom.c	Scalefactor band table
sfb_8_1024	short	41	aac_rom.c	Scalefactor band table
sfb_8_128	short	16	aac_rom.c	Scalefactor band table
HuffmanCodeBook_1	short	204	aac_rom.c	Huffman codeword table AAC
HuffmanCodeBook_2	short	156	aac_rom.c	Huffman codeword table AAC
HuffmanCodeBook_3	short	156	aac_rom.c	Huffman codeword table AAC
HuffmanCodeBook_4	short	152	aac_rom.c	Huffman codeword table AAC
HuffmanCodeBook_5	short	164	aac_rom.c	Huffman codeword table AAC
HuffmanCodeBook_6	short	160	aac_rom.c	Huffman codeword table AAC
HuffmanCodeBook_7	short	124	aac_rom.c	Huffman codeword table AAC
HuffmanCodeBook_8	short	124	aac_rom.c	Huffman codeword table AAC
HuffmanCodeBook_9	short	336	aac_rom.c	Huffman codeword table AAC
HuffmanCodeBook_10	short	328	aac_rom.c	Huffman codeword table AAC
HuffmanCodeBook_11	short	544	aac_rom.c	Huffman codeword table AAC
HuffmanCodeBook_SCL	short	260	aac_rom.c	Huffman codeword table AAC
SamplingRateInfoTable	mixed	45	aac_rom.c	Sampling rate to scalefactor mapping
				table AAC
HuffmanCodeBooks	mixed	52	aac_rom.c	Huffman codeword table AAC
tns_max_bands_tbl	char	18	aac_rom.c	max. TNS bands per sampling rate table
sbr_limGains	float	4	sbr_rom.c	SBR limiter gain values
sbr_limiterBandsPerOctave	float	4	sbr_rom.c	Number of SBR limiter bands
sbr_smoothFilter	float	4	sbr_rom.c	Smoothing filter for gain values
sbr_invIntTable	float	55	sbr_rom.c	Table of 1/x function
sbr_randomPhase	float	1024	sbr_rom.c	Random numbers for SBR noise addition
				and PNS
sbr_qmf_64_640	float	325	sbr_rom.c	QMF window coefficients
sbr_cos_twiddle_L04	float	2	sbr_rom.c	FFT twiddle table
sbr_cos_twiddle_L08	float	4	sbr_rom.c	FFT twiddle table
sbr_cos_twiddle_L16	float	8	sbr_rom.c	FFT twiddle table
sbr_cos_twiddle_L32	float	16	sbr_rom.c	FFT twiddle table
sbr_sin_twiddle_L04	float	2	sbr_rom.c	FFT twiddle table
sbr_sin_twiddle_L08	float	4	sbr_rom.c	FFT twiddle table
sbr_sin_twiddle_L16	float	8	sbr_rom.c	FFT twiddle table
sbr_sin_twiddle_L32	float	16	sbr_rom.c	FFT twiddle table
sbr_alt_sin_twiddle_L04	float	3	sbr_rom.c	FFT twiddle table
sbr_alt_sin_twiddle_L08	float	5	sbr_rom.c	FFT twiddle table
sbr_alt_sin_twiddle_L16	float	9	sbr_rom.c	FFT twiddle table
sbr_alt_sin_twiddle_L32	float	17	sbr_rom.c	FFT twiddle table
sbr_cos_twiddle_ds_L32	float	32	sbr_rom.c	FFT twiddle table, obsolete for mono only
				decoder
sbr_sin_twiddle_ds_L32	float	32	sbr_rom.c	FFT twiddle table, obsolete for mono only
				decoder
sbr_cos_twiddle_L64	float	32	sbr_rom.c	FFT twiddle table, obsolete for mono only
				decoder
sbr_sin_twiddle_L64	float	32	sbr_rom.c	FFT twiddle table, obsolete for mono only
				decoder
sbr_alt_sin_twiddle_L64	float	33	sbr_rom.c	FFT twiddle table, obsolete for mono only
				decoder
sbr_t_cos_L32	float	32	sbr_rom.c	FFT twiddle table
	float	32	sbr_rom.c	FFT twiddle table
sbr_t_sin_L32				
sbr_t_sin_L32 aRevLinkDecaySer	float	3	sbr_rom.c	Parametric Stereo all-pass filter
		3	sbr_rom.c	Parametric Stereo all-pass filter coefficients
		3 20	sbr_rom.c sbr_rom.c	

aFractDelayPhaseFactorReSubQmf	float	10	sbr_rom.c	Parametric Stereo phase rotation factor
aFractDelayPhaseFactorImSubQmf	float	10	sbr_rom.c	Parametric Stereo phase rotation factor
aaFractDelayPhaseFactorSerReQmf	float	3	sbr_rom.c	Parametric Stereo phase rotation factor
aaFractDelayPhaseFactorSerImQmf	float	3	sbr_rom.c	Parametric Stereo phase rotation factor
aaFractDelayPhaseFactorSerReSubQmf	float	3	sbr_rom.c	Parametric Stereo phase rotation factor
aaFractDelayPhaseFactorSerImSubQmf	float	3	sbr_rom.c	Parametric Stereo phase rotation factor
scaleFactors	float	15	sbr_rom.c	Parametric Stereo quantization table
scaleFactorsFine	float	41	sbr_rom.c	Parametric Stereo quantization table
alphas	float	8	sbr_rom.c	Parametric Stereo quantization table
p2_6	float	6	sbr_rom.c	Hybrid filterbank coefficients
p8_13	float	14	sbr_rom.c	Hybrid filterbank coefficients
sbr whFactorsTable	float	54	sbr_rom.c	Tuning parameters for inverse filtering
bins2groupMap	short	22	sbr_rom.c	Mapping of Parametric Stereo bins to
biliszgroupiviap	SHOIL	22	SDI_10111.C	Parametric Stereo groups
sbr_whFactorsIndex	short	9	sbr_rom.c	Tuning parameter index for inverse
SDI_WIIFACIOISIIIQEX	SHOIL	9	SDI_10111.C	filtering
sbr_start_freq_16	char	16	sbr_rom.c	SBR frequency scale index
sbr_start_freq_22	char	16	sbr_rom.c	SBR frequency scale index
sbr_start_freq_24	char	16	sbr_rom.c	SBR frequency scale index
sbr_start_freq_24	char	16	sbr_rom.c	SBR frequency scale index
sbr_start_freq_44	char	16	sbr_rom.c	SBR frequency scale index
sbr_start_freq_48	char	16	sbr_rom.c	SBR frequency scale index
sbr_frame_info1_16	char	18	sbr_rom.c	SBR frequency scale index
sbr_frame_info2_16	char	18	sbr_rom.c	SBR frequency scale index
sbr_frame_info4_16	char	18	sbr_rom.c	SBR frequency scale index
sbr_huffBook_EnvLevel10T	char	240	sbr_rom.c	Huffman codeword table SBR
sbr_huffBook_EnvLevel10F	char	240	sbr_rom.c	Huffman codeword table SBR
sbr_huffBook_EnvBalance10T	char	96	sbr_rom.c	Huffman codeword table SBR
sbr_huffBook_EnvBalance10F	char	96	sbr_rom.c	Huffman codeword table SBR
sbr_huffBook_EnvLevel11T	char	124	sbr_rom.c	Huffman codeword table SBR
sbr_huffBook_EnvLevel11F	char	124	sbr_rom.c	Huffman codeword table SBR
sbr_huffBook_EnvBalance11T	char	48	sbr_rom.c	Huffman codeword table SBR
sbr_huffBook_EnvBalance11F	char	48	sbr_rom.c	Huffman codeword table SBR
sbr_huffBook_NoiseLevel11T	char	124	sbr_rom.c	Huffman codeword table SBR
sbr_huffBook_NoiseBalance11T	char	48	sbr_rom.c	Huffman codeword table SBR
aRevLinkDelaySer	char	3	sbr_rom.c	Parametric Stereo all-pass delay line
artevening oray cor	oriai	Ŭ	001_1011110	lengths
groupBorders	char	23	sbr_rom.c	Borders of Parametric Stereo groups
aBookPslidTimeDecode	char	56	sbr_rom.c	Huffman codeword table Parametric
				Stereo
aBookPslidFreqDecode	char	56	sbr_rom.c	Huffman codeword table Parametric
				Stereo
aBookPsIccTimeDecode	char	28	sbr_rom.c	Huffman codeword table Parametric
			_	Stereo
aBookPslccFreqDecode	char	28	sbr_rom.c	Huffman codeword table Parametric
				Stereo
aBookPslidFineTimeDecode	char	120	sbr_rom.c	Huffman codeword table Parametric
				Stereo
aBookPslidFineFreqDecode	char	120	sbr_rom.c	Huffman codeword table Parametric
•				Stereo
sbr_defaultHeader	char	32	sbr_rom.c	Default SBR header data
logDualisTable	float	65	transcendent.c	Lookup table for efficient log()
				implementation
Sum		9866		

4.3.2 Static memory

This clause contains a listing of all static buffers contributing to the RAM requirements of the encoder and decoder.

Table 9: Encoder static memory

Name	Data type	Size	Allocated in	Description
		[word]	Source File	
mdctDelayBuffer	float	3200	aac_ram.c	Time domain input signal delay
sideInfoTabLong	int	52	aac_ram.c	Table lookup for side information, long
				blocks
sideInfoTabShort	int	16	aac_ram.c	Table lookup for side information, short
				blocks
aacEncoder	AAC_ENCODER	3554	aacenc.c	AAC encoder instance
sbr_QmfStatesAnalysis	float	1280	sbr_ram.c	QMF filterbank states buffer
sbr_envYBuffer	float	4096	sbr_ram.c	QMF band energy buffer
sbr_quotaMatrix	float	512	sbr_ram.c	Tonality values
sbr_thresholds	float	128	sbr_ram.c	Detector parameters
sbr_toncorrBuff	float	1256	sbr_ram.c	Detector value buffer
EnvChannel[nChan]	ENV_CHANNEL	1794	sbr_main.c	SBR channel instance, only half the
				size for mono only encoder
sbrEncoder	SBR_ENCODER	200	sbr_main.c	SBR encoder instance
SynthesisQmfBank	SBR_QMF_FILTE	7	sbr_main.c	QMF synthesis filterbank instance
	R_BANK			
psEncoder	PS_ENC	281	sbr_main.c	Parametric Stereo encoder instance
sbr_freqBandTableLO	char	14	sbr_ram.c	SBR frequency band table, low
				resolution
sbr_freqBandTableHI	char	28	sbr_ram.c	SBR frequency band table, high
				resolution
sbr_v_k_master	char	28	sbr_ram.c	SBR frequency band table index
sbr_guideScfb	char	54	sbr_ram.c	Additional sine detection parameter
sbr_detectionVectors	char	216	sbr_ram.c	Additional sine detection parameter
sbr_prevEnvelopeCompensa	char	54	sbr_ram.c	Additional sine detection parameter
tion				
sbr_guideVectorDetected	char	216	sbr_ram.c	Additional sine detection parameter
outputBuffer	int	384	main.c	Bitstream output buffer
inputBuffer[nChan]	float	7202	main.c	Time domain input signal buffer, only
				half the size for mono only encoder
IIR21_resampler[nChan]	float	144	main.c	2:1 IIR resampler instance (includes
				states), only half the size for mono only
				encoder
statesIIR	float	16	iir32resample	3:2 IIR resampler states buffer
		0.480	.C	
Sum		24732		

Table 10: Decoder static memory

Name	Data type	Size	Allocated in	Description
		[word]	Source File	
OverlapBuffer[nChan]	float	1024	aac_ram.c	Delay buffer for overlap and add, only half the size for mono only decoder
AacDecoderInstance	AAC_DECODER_INS TANCE	11	aacdecoder.c	AAC decoder instance
StreamInfo	CStreamInfo	7	aac_ram.c	Bitstream information
AacDecoderStaticChannelInfo[nChan]	CaacDecoderStaticCh annelInfo	16	aac_ram.c	Channel information, only half the size for mono only decoder
sbr_CodecQmfStatesAnalysis	float	640	sbr_ram.c	QMF analysis filter bank states
sbr_GainSmooth	float	96	sbr_ram.c	Gain smoothing filter states
sbr_NoiseSmooth	float	96	sbr_ram.c	Noise level smoothing filter states
sbr_QmfStatesSynthesis	float	1280	sbr_ram.c	QMF synthesis filter bank states
sbr_OverlapBuffer	float	1536	sbr_ram.c	SBR delay buffer, only half the size for mono only decoder
sbr_LpcFilterStatesReal	float	128	sbr_ram.c	LPC filter states
sbr_LpcFilterStatesImag	float	128	sbr_ram.c	LPC filter states, obsolete for mono only decoder
sbr_TransposerSettings	float	18	sbr_ram.c	Transposer configuration parameters
FreqBandData	FREQ_BAND_DATA	164	sbr_ram.c	SBR Frequency band information
PrevFrameData[nChan]	SBR_PREV_FRAME_ DATA	120	sbr_ram.c	SBR previous frame data, only half the size for mono only decoder
sbr_PrevBitstream	SBRBITSTREAM	146	sbr_ram.c	SBR previous frame bitstream
sbrDecoderInstance	SBR_DECODER_INS TANCE	797	sbrdecoder.c	SBR decoder instance
TimeDataFloat[nChan]	float	4096	main.c	Output buffer for time-domain signal, only half the size for mono only decoder
inBuffer	int	384	main.c	Input buffer for bitstream
splineResamplerInstance	SPLINE_RESAMPLE R	21	spline_resam pler.c	Spline resampler instance
Sum		10708		

4.3.3 Dynamic memory

This clause contains a listing of all dynamic buffers contributing to the RAM requirements of the encoder and decoder. Dynamic memory can be re-used outside of the encoder or decoder application.

Table 11: Encoder dynamic memory

Name	Data	Size	Allocated in	Description
	type	[word]	Source File	-
PsBuf3	float	1024	sbr_ram.c	Note: reused in AAC encoder
sbr_envRBuffer	float	4096	sbr_ram.c	Note: reused in AAC encoder
sbr_envlBuffer	float	4096	sbr_ram.c	Note: reused in AAC encoder
sbr_transients	float	192	sbr_ram.c	Note: reused in AAC encoder
Sum		9408		

Table 12: Decoder dynamic memory

Name	Data	Size	Allocated in	Description
	type	[word]	Source File	
WorkBufferCore	float	2048	aac_ram.c	Note: reused in SBR decoder
InterimResult	float	1024	sbr_ram.c	
Sum		3072		

4.3.4 Maximum stack size

This clause contains tables for the encoder and the decoder which describe the call stack that results in the maximum stack size usage.

Table 13: Encoder call stack

Function	Local variables	Stack
		used
	aturat config.	[bytes]
main	struct config; int error;	20 4
	int error,	4
	int bitrate:	4
	int nChannelsAAC, nChannelsSBR;	8
	int sampleRateAAC;	4
	int bandwidth;	4
	unsigned int numAncDataBytes;	4
	unsigned char ancDataBytes[256];	256
	unsigned int ancDataLength;	4
	int numSamplesRead;	4
	int bDollR2Downsample;	4
	int bDingleRate;	4
	int useParametricStereo;	4
	int coreWriteOffset;	4
	int coreReadOffset;	4
	int envWriteOffset;	4
	int envReadOffset;	4
	int writeOffset;	4
	struct *aacEnc;	4
	int bDoUpsample;	4
	int upsampleReadOffset;	4
	int inSamples;	4
	int bDollR32Resample;	4
	int nSamplesPerChannel;	4
	const int nRuns;	4
	float *resamplerScratch;	4
	struct *hEnvEnc;	4
	int i, ch, outSamples, numOutBytes;	16 = 400
EnvEncodeFrame	struct *hEnvEncoder;	4
LIN LINGUIGH TAING	float *samples;	4
	float *pCoreBuffer;	4
	unsigned int timeInStride;	4
	unsigned int *numAncBytes;	4
	unsigned char *ancData;	4
	struct *sbrBitstreamData;	4
		= 28
extractSbrEnvelope	float *timeInPtr;	4
	float *pCoreBuffer;	4
	unsigned int timeInStride;	4
	struct *h_con;	4
	struct *sbrHeaderData;	4
	struct *sbrBitstreamData;	4
	struct *h_envChan[];	4
	struct *h_ps_e;	4
	struct *hSynthesisQmfBank;	4
	struct *hCmonData;	4
	int ch, i, j, c;	16
	int nEnvelopes[2];	8
	int transient_info[2][2];	16 g
	const struct *frame_info[2];	8
	int nChannels, nInChannels;	8
	enum stereoMode; enum res[10];	4 40
	enum res[10]; int v_tuning[6];	24
	int v_tuning[o], int sfb_nrg [2][135];	1080
1	float noiseFloor[2][10];	80
	moat noison ioon[2][10],	1 00

	<u></u>	
	int noise_level[2][10];	80
	int sfb_nrg_coupling[2][135];	1080
	int noise_level_coupling[2][10];	80
	int maxQuantError;	4
	int maxedantenor,	= 2568
EncodePsFrame	struct *pms;	4
Ziloodor or famo	float **iBufferLeft,	4
	float **rBufferLeft,	4
	float **iBufferRight,	4
	float **rBufferRight	4
	int env, i, bin, subband, maxSubband, startSample, stopSample;	28
	float **hybrLeftImag, **hybrLeftReal, **hybrRightImag, **hybrRightReal;	16
		= 64
HybridAnalysis	const float **mQmfReal;	4
,	const float **mQmfImag;	4
	float **mHybridReal;	4
	float **mHybridImag;	4
	struct *hHybrid;	4
	int n, band;	8
	enum hybridRes;	4
	int chOffset;	4
		= 36
eightChannelFiltering	const float *pQmfReal;	4
	const float *pQmflmag;	4
	float **mHybridReal;	4
	float **mHybridImag;	4
	int i, n;	8
	float real, imag;	8
	int midTap;	4
	float cum[16];	64
	moat outilitoj,	= 100
CFFTN	float *afftData;	4
CITIN	int len;	_
		4
	int isign;	4
***	(, , D , D	= 12
cfftn	float Re[];	4
	float Im[];	4
	int nTotal;	4
	int nPass;	4
	int nSpan;	4
	int iSign;	4
	int ii, mfactor, kspan, ispan, inc, j, jc, jf, jj, k, k1, k2, k3, k4, kk, kt, nn, ns, nt;	76
	double radf, c1, c2, c3, cd, s1, s2, s3, sd;	72
	float ak, bk, akp, bkp, ajp, bjp, ajm, bjm, akm, bkm, aj, bj, aa, bb;	56
	float Rtmp[23], Itmp[23];	184
	double Cos[23], Sin[23];	368
	int Perm[209];	836
	int Ferril[209],	
		44
	double s60, c72, s72, pi2;	32
		= 1692
	Sum	4900

Table 14: Decoder call stack

Function	Local variables	Stack used
		[bytes]
main()	int endOfFile;	4
	char frameOk;	1
	int i; int written16;	4 4
	char channelMode;	1
	struct *hBitBuf;	4
	struct *aacDecoderInfo;	4
	struct *streamSBR;	576
	struct *sbrDecoderInfo;	4
	struct * splineResampler;	4
	int frameSize;	4
	int sampleRate, outputSampleRate;	8
	int numChannels;	4
	int numOutSamples; int bDownSample;	4 4
	int fosr16, fosr8;	8
	int bBitstreamDownMix;	4
	int bValidMode;	4
	in branchists,	= 646
applySBR()	struct *self;	4
	struct *Bitstr;	4
	float *timeData;	4
	int *numChannels;	4
	int SbrFrameOK;	4
	int bDownSample;	4
	int bBitstreamDownMix;	4
	unsigned char i, dualMono; int stereo, CRCLen, crcEnable, readHeader, err;	2 20
	struct *SbrChannel;	4
	struct bitBuf;	16
	struct *hHeaderData;	4
	enum headerStatus;	4
	int codecFrameSize;	4
	enum initialSyncState;	4
	struct *hConcealData;	4
	float *pWorkBuffer1;	4
	struct *hFrameDataLeft; struct *hFrameDataRight;	4 4
	Struct HriameDataRight,	= 102
sbr_dec()	struct *hSbrDec;	4
0000()	float *timeIn;	4
	float *timeOut;	4
	float *interimResult;	4
	struct *hHeaderData;	4
	struct *hFrameData;	4
	struct *hPrevFrameData;	4
	int applyProcessing;	4
	struct *h_ps_d; struct *hSynthesisQmfBankRight;	4 4
	int nChannels;	4
	int i, k, slot, ov_len, bUseLP;	20
	float *QmfBufferReal[38];	152
	float *QmfBufferImag[38];	152
	float *ptr;	4
	int noCols, halflen, islots;	12
		= 384

10 11 10 1511 1 0	[4
cplxSynthesisQmfFiltering()		4
	float **qmflmag;	4
	float *timeout;	4
	struct *synQmf;	4
	int bUseLP;	4
	struct *h_ps_dec;	4
	int active;	4
	int i, j;	8
	float *ptr_time_out, *filterStates;	8
	float accu;	4
	int p;	4
	float qmfReal2[64];	256
	float *imagSlot;	4
	int no_synthesis_channels;	4
	int qmf_filter_state_syn_size;	4
		-
	float mfRealTmp[64];	256
	float qmflmagTmp[64];	256
	int env;	4
	const float *p_filter;	4
		= 840
ApplyPsSlot()	struct *h_ps_dec;	4
	float **rIntBufferLeft;	4
	float **iIntBufferLeft;	4
	float *rIntBufferRight;	4
	float *iIntBufferRight;	4
		= 20
HybridAnalysis()	const float **mQmfReal;	4
	const float **mQmflmag;	4
	float **mHybridReal;	4
	float **mHybridImag;	4
	struct *hHybrid;	4
	int n, band;	8
	enum hybridRes;	4
	int chOffset;	4
		= 36
eightChannelFiltering()	const float *pQmfReal;	4
30	const float *pQmflmag;	4
	float **mHybridReal;	4
	float **mHybridImag;	4
	int i, n;	8
	float real, imag;	8
	int midTap;	4
	float cum[16];	64
		= 100
CFFTN()	float *afftData;	4
011111()	int len;	4
	· ·	4
	int isign;	
		= 12
cfftn()	float Re[];	4
	float Im[];	4
	int nTotal;	4
	int nPass;	4
	int nSpan;	4
	int iSign;	4
	int ii, mfactor, kspan, ispan, inc, j, jc, jf, jj, k, k1, k2, k3, k4, kk, kt, nn,	76
	ns, nt;	72
	double radf, c1, c2, c3, cd, s1, s2, s3, sd;	56
	float ak, bk, akp, bkp, ajp, bjp, ajm, bjm, akm, bkm, aj, bj, aa, bb;	184
	float Rtmp[23], ltmp[23];	368
	double Cos[23], Sin[23];	836
	int Perm[209];	44
	int factor [11];	32
	double s60, c72, s72, pi2;	= 1692
	Sum	3832

5 File formats

This clause describes the file formats used by the encoder and decoder programs.

5.1 Audio input file (encoder input/decoder output)

The audio input files read by the encoder and written by the decoder are 16-bit PCM wave files. For convenient handling of wave files a precompiled audio-fileformat library is used.

5.2 Bitstream file format (encoder output/decoder input)

The encoder program writes and the decoder program reads raw frames packetized in access units as described by 3GPP TS 26.244. For packetization the ISO media library is used. A precompiled library is used.

5.3 Error pattern file (decoder input)

The decoder program can optionally process an additional input file which describes an error pattern. The format of the error pattern file is 1 character per line. Each line corresponds to one frame, where a '0' indicates that the respective frame has been transmitted without errors, while a '1' indicates that the corresponding frame has been lost and error concealment shall be applied by the decoder.

Annex A (informative): Weighted MOPS and PROM

The complexity numbers for the Enhanced aacPlus audio codec can be found in the following table, the numbers have been derived using the 'allcat.wav' item, which holds all the material from the selection test concatenated in one single item. For every test case the average and worst frame weighted MOPS figure has been derived. The worst case wMOPS figure over all test cases has been marked in **blue**.

Table A.1: Weighted MOPS and PROM figures

	Test Case	Mono Encoder	Stereo Encoder	Decoder	Decoder, mono only
	14m	15.23 / 16.98	15.36 / 17.21	9.38 / 10.07	8.07 / 8.78
	18s		25.79 / 28.36	19.48 / 20.35	8.31 / 9.17
	24m	16.72 / 18.93	16.86 / 19.14	10.30 / 11.39	8.89 / 9.94
wMOPS	24s		27.01 / 29.85	20.45 / 21.63	8.82 / 9.93
	32s		27.49 / 29.97	21.08 / 22.42	9.28 / 10.58
[average / worst frame]	48s		35.22 / 42.22	17.96 / 20.26	12.42 / 14.32
worst framej	14m, 16 kHz	15.42 / 18.41	15.47 / 18.46	7.85 / 8.61	7.85 / 8.60
	14m, 3% FER			9.38 / 10.07	8.07 / 8.78
	24s, 3% FER			20.45 / 21.63	8.81 / 9.93
	32s, 1%FER			21.08 / 22.42	9.28 / 10.58
	32s, 3%FER			21.08 / 22.38	9.27 / 10.58
Program ROM [ops]		12540	14365	8048	6209

Annex B (informative): Change history

Det	T00 01"	TCC D	CD	D -	Change history	01.1	Name
Date	TSG SA#	TSG Doc.	CR	Rev	Subject/Comment	Old	New
2004-09	25	SP-040638	201		Approved at SA#25	2.0.0	6.0.0
2004-12	26	SP-040840			Correction to C-code to increase error robustness	6.0.0	6.1.0
2004-12	26	SP-040840			Correction to C-code: Missing memory re-initialization	6.0.0	6.1.0
2004-12	26	SP-040840			Correction to C-code: Memory initialization added	6.0.0	6.1.0
2004-12	26	SP-040840			Correction to C-code: Wrong calculation of sine levels	6.0.0	6.1.0
2004-12	26	SP-040840	005		Correction to C-code: Prevent multiple reading of	6.0.0	6.1.0
000110		00.010010	200		bitstream elements	0.00	0.4.0
2004-12	26	SP-040840		2	Correction to C-code: Corrected wrong table values	6.0.0	6.1.0
2004-12	26	SP-040840			Correction to C-code: Modify instrumentation	6.0.0	6.1.0
2004-12	26	SP-040840	800	1	Correction of C-code: Output data was copied into	6.0.0	6.1.0
000110		00.040040	000		wrong array	0.00	0.4.0
2004-12	26	SP-040840		1	Correction to C-code: Bug in resampler	6.0.0	6.1.0
2004-12	26	SP-040840		1	Correction to C-code: Modify data types for FFT	6.0.0	6.1.0
2004-12	26	SP-040840		1	Correction to decoder C-Code: Alignment with MPEG specification	6.0.0	6.1.0
2004-12	26	SP-040840	012		Correction to C-code: Reset of Missing Harmonics flags during concealment added	6.0.0	6.1.0
2004-12	26	SP-040840	013		Removal of Complexity counters	6.0.0	6.1.0
2005-01					File "env_calc.c" replaced in the attached ANSI-C code	6.1.0	6.1.1
2005-03	27	SP-050095	014	1	Correction to C-code: 3GPP file format wrong writing of brand	6.1.1	6.2.0
2005-03	27	SP-050095	015	1	Correction to C-code: remove copyright notice from 3GPP file format header files	6.1.1	6.2.0
2005-03	27	SP-050095	016	1	Correction to C-code: add capability for 10 kbit/s, mono encoding	6.1.1	6.2.0
2005-03	27	SP-050095	017	1	Correction to C-code: add capability for data stream element parsing	6.1.1	6.2.0
2005-03	27	SP-050095	018	1	Correction to C-code: PNS decoding algorithm not conform to MPEG	6.1.1	6.2.0
2005-03	27	SP-050095	019	1	Correction to C-code: the decoder mono only compile target not working correctly	6.1.1	6.2.0
2005-03	27	SP-050095	020	1	Correction to C-code: PS-decoding with varying upper frequency border not working correctly	6.1.1	6.2.0
2005-03	27	SP-050095	021	1	Correction to C-code: PS-decoding with variable framing not working correctly	6.1.1	6.2.0
2005-03	27	SP-050095	022		Correction to written specification: move WMOPS numbers to informative Annex	6.1.1	6.2.0
2005-06	28	SP-050251	023		Correction to C-code: 10 kbit/s mono encoding with stereo input files failed	6.2.0	6.3.0
2005-09	29	SP-050426	0024		Correction to C-Code: Enable 44.1 kHz input material encoding (floating-point code)	6.3.0	6.4.0
2005-09	29	SP-050426	0025		Correction of C-Code: removal of obsolete table (floating-point code)	6.3.0	6.4.0
2005-12	30	SP-050786	0027		Correction to C-code: encoder bitrate switching simulation toolset	6.4.0	6.5.0
2005-12	30	SP-050786	0028		Correction to C-code: encoder tuning table entry for 44.1 kHz was wrong	6.4.0	6.5.0
2005-12	30	SP-050786	0029		Correction to C-code: fix to make decoder more robust against corrupt input data	6.4.0	6.5.0
2005-12	30	SP-050786	0030		Correction to C-code: removal of unused coefficients in resampler	6.4.0	6.5.0
2006-03	31	SP-060013	0031		Correction to C-code: encoder switch is wrong	6.5.0	6.6.0
2006-09	33	SP-060595		1	Correction of written specification: correct memory tables (floating-point code)	6.6.0	6.7.0
2006-09	33	SP-060601	0032	2	Modification of C-code: Change of encoder bitrate	6.7.0	7.0.0
					border for Parametric Stereo usage (floating-point code)		

2006-12	34	SP-060849	0035		Correction to C-code: Correct the maximum possible SBR payload size (floating point code)	7.0.0	7.1.0
2007-09	37	SP-070628	0038	1	Adding safety code to avoid encoder runtime-assertion	7.1.0	7.2.0
2008-12	42				Version for Release 8	7.2.0	8.0.0
2009-12	46				Version for Release 9	8.0.0	9.0.0
2011-03	51				Version for Release 10	9.0.0	10.0.0

History

Document history		
V10.0.0	April 2011	Publication