ETSITS 126 304 V6.5.0 (2006-03)

Technical Specification

Digital cellular telecommunications system (Phase 2+); Universal Mobile Telecommunications System (UMTS); Extended Adaptive Multi-Rate - Wideband (AMR-WB+) codec; Floating-point ANSI-C code (3GPP TS 26.304 version 6.5.0 Release 6)



Reference
RTS/TSGS-0426304v650

Keywords
GSM, 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

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 2006.
All rights reserved.

DECTTM, **PLUGTESTS**TM and **UMTS**TM are Trade Marks of ETSI registered for the benefit of its Members. **TIPHON**TM and the **TIPHON logo** are Trade Marks currently being registered by ETSI for the benefit of its Members. **3GPP**TM is a Trade Mark of ETSI registered for the benefit of its Members and of the 3GPP Organizational Partners.

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

Intell	ectual Property Rights	2
	word	
Forev	word	4
1	Scope	5
2	References	
3	Definitions, symbols and abbreviations	5
3.1	Definitions	
3.2	Abbreviations	
4	C code structure	6
4.1	Contents of the C source code	6
4.2	Program execution	6
4.3	Code hierarchy	6
4.4	Variables, constants and tables	19
4.4.1	Description of fixed tables used in the C-code	19
4.4.2	Static variables used in the C-code	20
5	File formats	24
5.1	Audio file (encoder input/decoder output)	24
5.2	Parameter bitstream file (encoder output/decoder input)	
Anne	ex A (informative): Change history	26
Histo	ory	27
1	Л У	

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 Extended Adaptive Multi-Rate Wideband codec. Alternatively, fixed-point ANSI-C code is specified in 3GPP TS 26.273 [1]. The floating-point codec/encoder/decoder specified in this document or the fixed-point codec/encoder/decoder specified in [1] may be used depending on if the implementation platform is better suited for a floating-point or a fixed-point implementation. It has been verified that the fixed-point and floating-point codecs interoperate with each other without any artifacts.

The floating-point ANSI-C code in the present document defines, besides the fixed-point c-code specified in [1], one valid reference implementation of the Extended Adaptive Multi-Rate Wideband transcoder (3GPP TS 26.290 [2]). Standard conformance is enforced by meeting the conformance criteria defined in [3].

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 26.273: "ANSI-C code for the Fixed-point Extended AMR Wideband codec".
- [2] 3GPP TS 26.290: " Audio codec processing functions; Extended AMR Wideband codec; Transcoding functions ".
- [3] 3GPP TS 26.274: " Audio codec processing functions; Extended Adaptive Multi-Rate Wideband (AMR-WB+) codec; Conformance testing ".
- [4] 3GPP TS 26.244: "Transparent end-to-end packet switched streaming service (PSS); 3GPP file format (3GP)"

3 Definitions, symbols and abbreviations

3.1 Definitions

For the purposes of the present document, the terms and definitions are given in TS 26.290 [2].

3.2 Abbreviations

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

AMR-WB+ Extended Adaptive Multi-Rate WideBand ANSI American National Standards Institute GSM Global System for Mobile communications

I/O Input/Output

RAM Random Access Memory ROM Read Only Memory

4 C code structure

This clause gives an overview of the structure of the 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 2000 SP4 and Microsoft Visual C++ v.6.0 compiler.

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

4.1 Contents of the C source code

The C code distribution has the files divided in five different directories, all present in the directory *c-code*. The directories are: *common*, *decoder*, *encoder*, *lib_amr* and *include*. The distributed files with suffix "c" contain the source code and the files with suffix "h" are the header files.

Project and workspace files are provided in the directory MSVC.

4.2 Program execution

The Extended Adaptive Multi-Rate Wideband codec is implemented in two programs:

- (encoder) audio encoder;
- (decoder) audio decoder.

The programs should be called like:

- encoder [encoder options] -if <audio input file> -of <parameter file>;
- decoder [decoder options] –if <parameter file> -of <audio output file>.

The input files contain one or two channels of 16-bit linear encoded PCM audio samples stored in the wav file format and the parameter files contain encoded audio data and some additional flags.

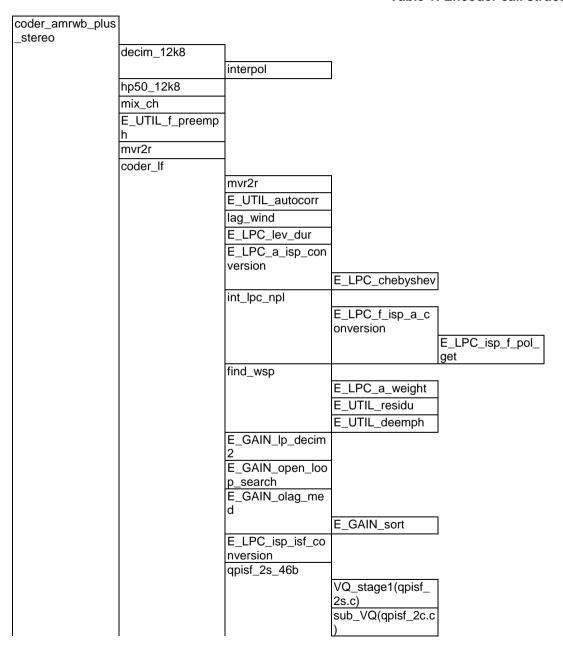
The encoder and decoder options will be explained by running the applications without input arguments. See the file readme.txt for more information on how to run the *encoder* and *decoder* programs.

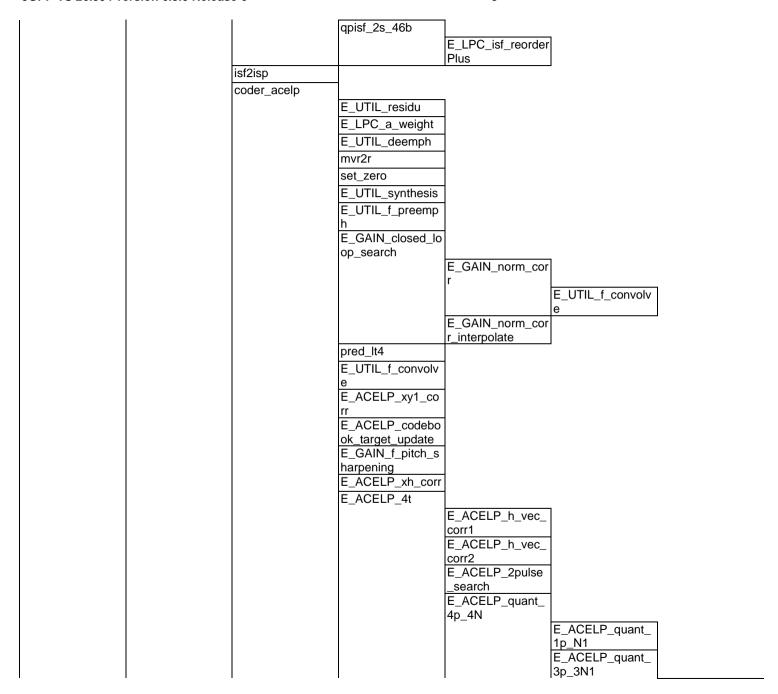
4.3 Code hierarchy

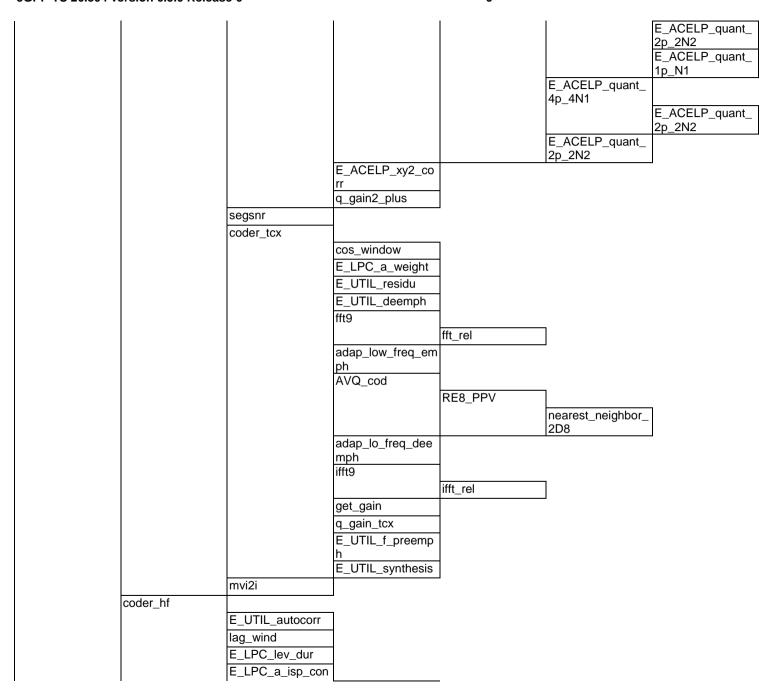
Tables 1 and 2 are call graphs that show the functions used in the audio codec.

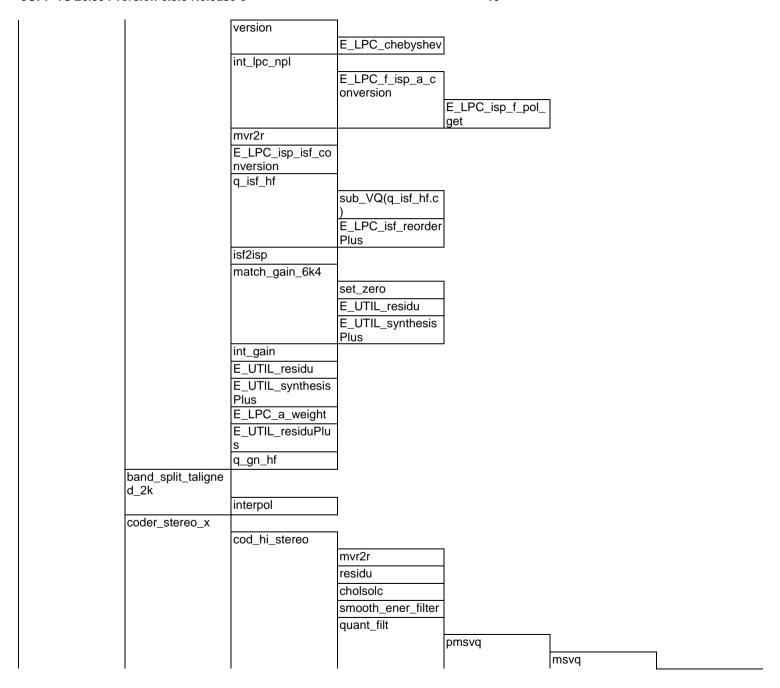
Each column represents a call level and each cell a function. The functions contain calls to the functions in rightwards neighbouring cells. The time order in the call graphs is from the top downwards as the processing of a frame advances. All standard C functions: memcpy(), fwrite(), etc. have been omitted. The initialization of the static RAM (i.e. calling the _init functions) is also omitted.

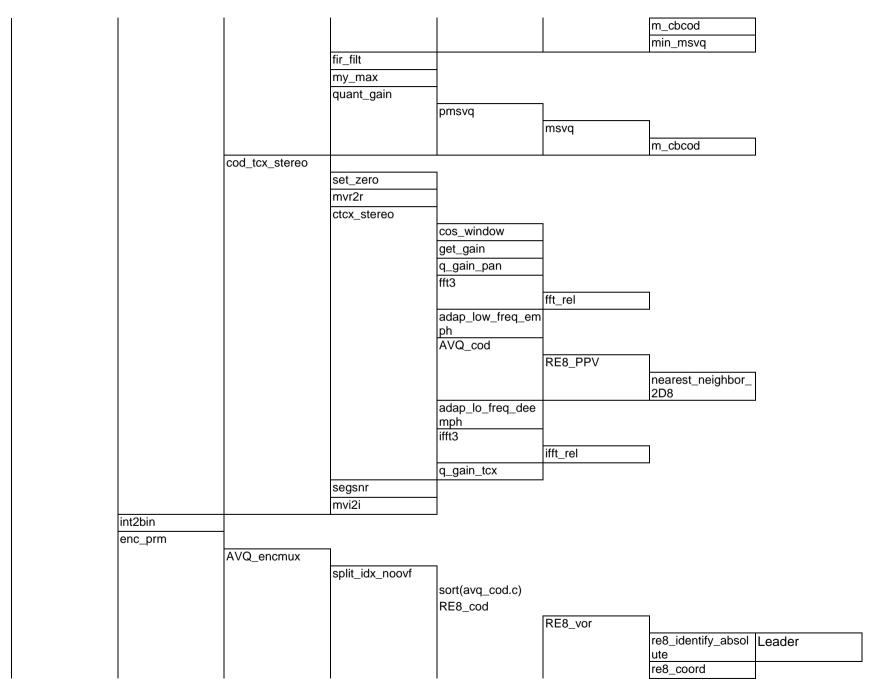
Table 1: Encoder call structure

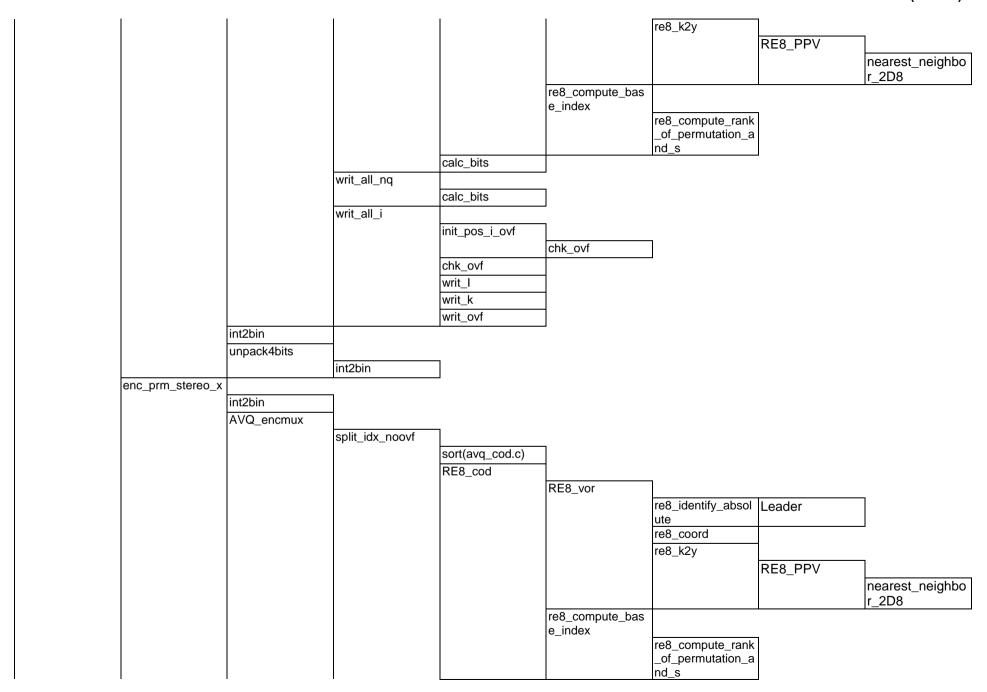












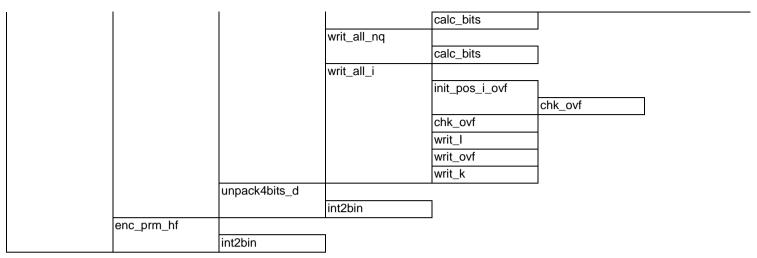
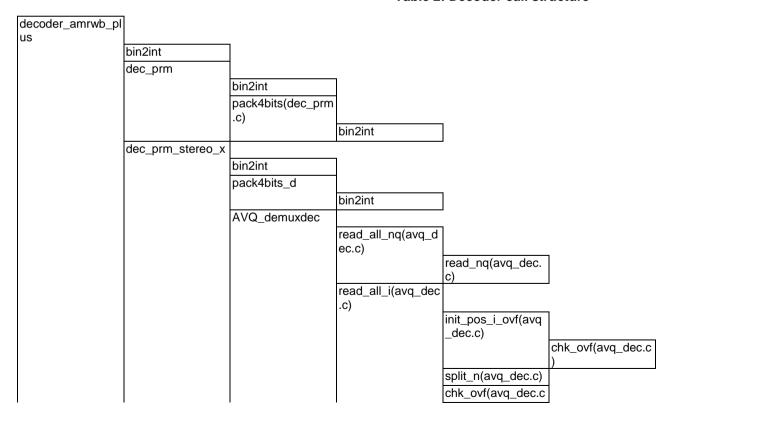
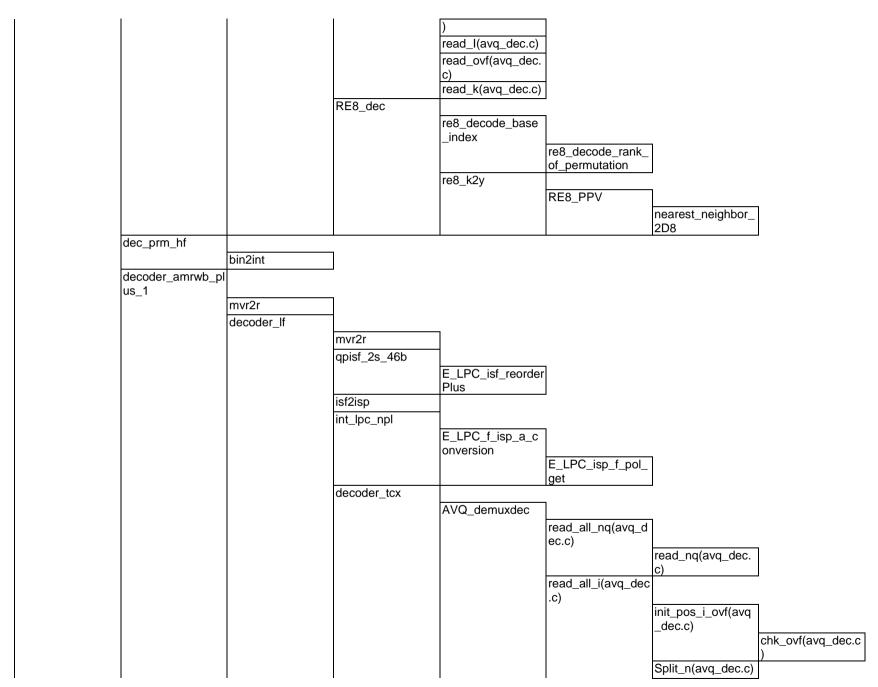
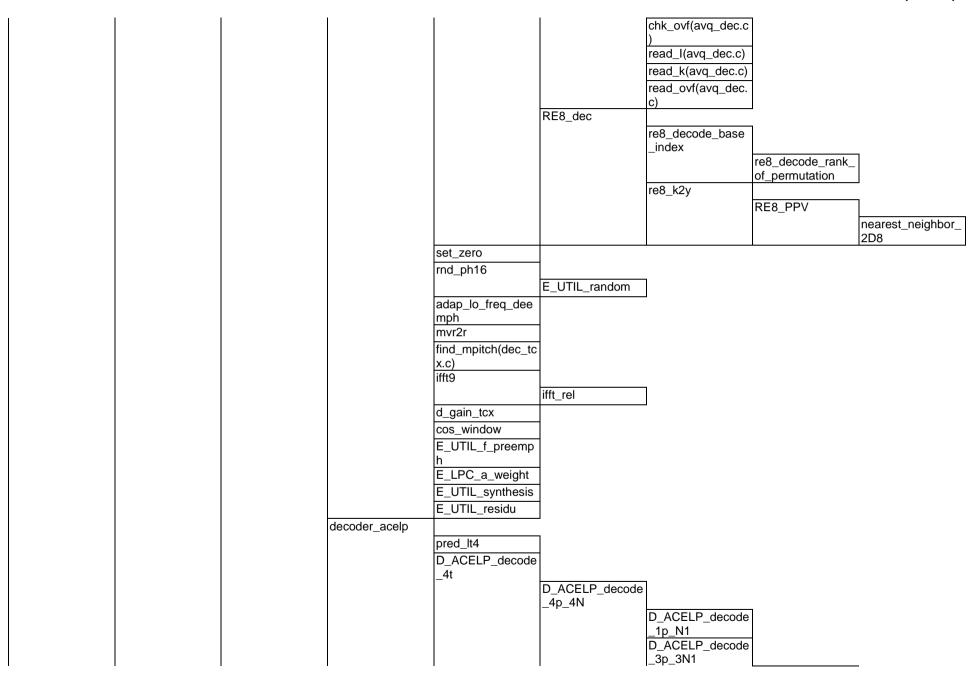
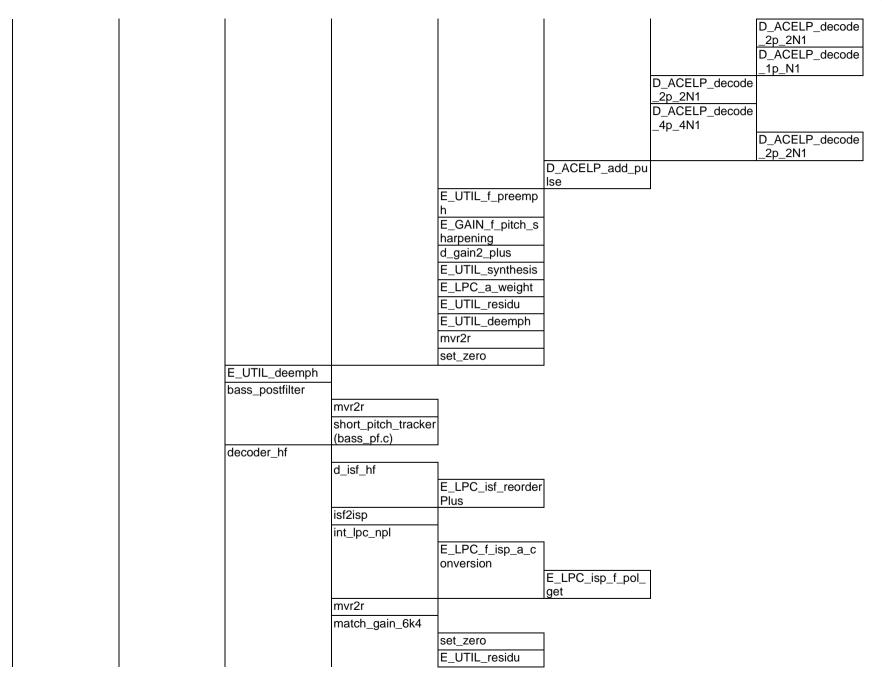


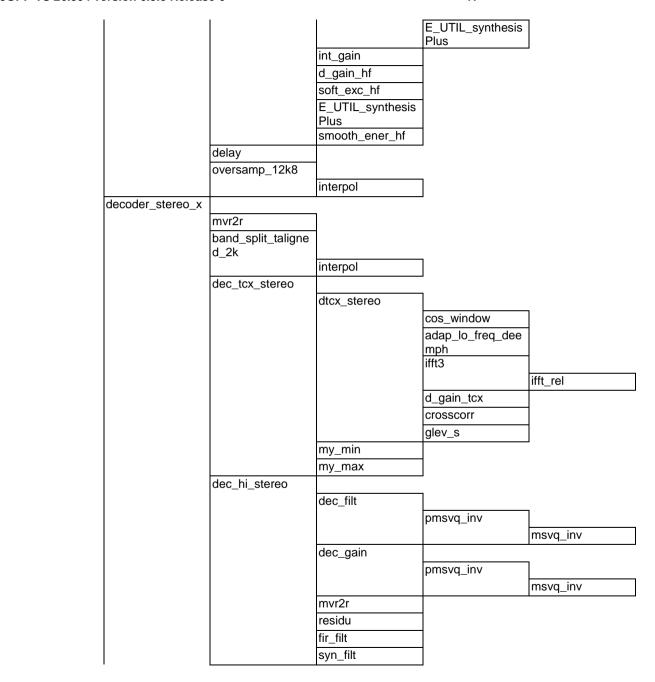
Table 2: Decoder call structure











	delay	
	band_join_2k	
		interpol
hp50_12k8		
oversamp_12k8		
	interpol	

4.4 Variables, constants and tables

4.4.1 Description of fixed tables used in the C-code

This clause contains a listing of all fixed tables declared in tables_plus.c and tables_stereo.c files.

Table 3: Encoder fixed tables

Float32	Format	Table name	Size	Description	
Fioral Fioral Fioral Fire Cosine table Fire Fire Cosine table Fire Fire	Float32	NBITS_CORE	8	Core bit-rates	
Float32	Float32	T_sin	1152	FFT Sine table	
Float32	Float32	T_cos	1152	FFT Cosine table	
Float32	Float32	filter_32k	61	FIR table for decimation/oversampling	
Float32	Float32	filter_32k_hf	61	FIR table for decimation/oversampling	
Fioat32	Float32	filter_32k_7k	61	FIR table for decimation/oversampling	
Float32	Float32	filter_48k	185	FIR table for decimation/oversampling	
Float32	Float32	Filter_48k_hf	185	FIR table for decimation/oversampling	
Float32	Float32	filter_8k	61	FIR table for decimation/oversampling	
Float32	Float32	isf_init	16	Initial ISF memory	
Float32	Float32	Mean_isf	16	Means of ISFs	
Float32	Float32	Dico1_isf	2304	1st stage codebook, isf0 to isf8	
Float32	Float32	Dico2_isf	1792	1st stage codebook, isf9 to isf15	
Float32	Float32	Dico21_isf	192	2nd stage codebook, isf2_0 to isf 2_2	
Float32	Float32	Dico22_isf	384	2nd stage codebook, isf2_3 to isf 2_5	
Float32	Float32	Dico23_isf	384	2nd stage codebook, isf2_6 to isf 2_8	
Float32	Float32	Dico24_isf	96	2nd stage codebook, isf2_9 to isf 2_11	
Float32	Float32	Dico25_isf	128	2nd stage codebook, isf2_12 to isf 2_15	
Float32	Float32	Dico21_isf_36b	640	1st stage codebook, (36b) split 1	
Float32 Dico_gain_hf 512 Quantization table for one-stage HF gain Float32 Mean_isf_hf_12k8 8 Means of ISFs (full band) Float32 dico1_isf_hf_12k8 32 1nd stage isf codebook (full band) Float32 mean_isf_hf_low_rate 8 Means of isfs Float32 Dico1_isf_hf_low_rate 32 1st stage isf codebook Float32 Dico1_isf_hf_low_rate 32 1st stage isf codebook Float32 Lag_window 17 Lag window Float32 Filt_lp 13 Low-pass fir filter for bass post filter Float32 Filt_lp 13 Low-pass fir filter for bass post filter Float32 Sin20 20 Random phase Float32 Inter4_2 65 ¼ resolution interpolation filter Float32 VadFiltBandFreqs 12 Open-loop classifier Float32 Bw 12 Open-loop classifier Float32 Lwg 8 Open-loop classifier Float32 Gain_hf_ramp 64 HF gain ramp for wb->wb+ switching Float32 Inter2_coef 12 Filter coefficients for band join/split Float32 Filter_LP180 2341 Filter for 48 kHz interpolation Float32 Filter_LP180 2341 Filter for 48 kHz interpolation Float32 Filter_2k 321 2k decimation filter Float32 Filt_hi_mscb4a 16*9 Float32 Filt_hi_mscb4a 16*9 Float32 Filt_hi_mscb_7a 16*9 Float32 Filt_hi_mscb_7a 16*9 Float32 Gain_hi_mean 2 Average gain vector Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2	Float32	Dico22_isf_36b	512	1st stage codebook, (36b) split 2	
Float32	Float32	Dico23_isf_36b	448	1st stage codebook, (36b) split 3	
Float32	Float32	Dico_gain_hf	512	Quantization table for one-stage HF gain	
Float32	Float32	Mean_isf_hf_12k8	8	Means of ISFs (full band)	
Float32	Float32	dico1_isf_hf_12k8	32	1nd stage isf codebook (full band)	
Float32	Float32	mean_isf_hf_low_rate		Means of isfs	
Float32 Lag_window 17 Lag window Float32 Filt_Ip 13 Low-pass fir filter for bass post filter Float32 Sin20 20 Random phase Float32 Inter4_2 65 ½ resolution interpolation filter Float32 VadFiltBandFreqs 12 Open-loop classifier Float32 Bw 12 Open-loop classifier Float32 Lwg 8 Open-loop classifier Float32 Lwg 8 Open-loop classifier Float32 Gain_hf_ramp 64 HF gain ramp for wb->wb+ switching Float32 Inter2_coef 12 Filter coefficients for band join/split Float32 Filter_LP180 2341 Filter for 48 kHz interpolation Float32 StereoNbits 18 Stereo bit-rates Float32 Filter_2k 321 2k decimation filter Float32 Filt_hi_mean 9 Average filter Float32 Filt_hi_mscb_7a 16*9 Float32 Filt_hi_mscb_7b 8*9 <td></td> <td>Dico1_isf_hf_low_rate</td> <td>32</td> <td>1st stage isf codebook</td>		Dico1_isf_hf_low_rate	32	1st stage isf codebook	
Float32 Filt_lp 13 Low-pass fir filter for bass post filter Float32 Sin20 20 Random phase Float32 Inter4_2 65 ¼ resolution interpolation filter Float32 VadFiltBandFreqs 12 Open-loop classifier Float32 Bw 12 Open-loop classifier Float32 Lwg 8 Open-loop classifier Float32 Gain_hf_ramp 64 HF gain ramp for wb->wb+ switching Float32 Inter2_coef 12 Filter coefficients for band join/split Float32 Filter_LP180 2341 Filter for 48 kHz interpolation Float32 StereoNbits 18 Stereo bit-rates Float32 Filter_2k 321 2k decimation filter Float32 Filter_brace 4 16*9 Float32 Filt_hi_mscb4a 16*9 Float32 Filt_hi_mscb_7a 16*9 Float32 Filt_hi_mscb_7b 8*9 Float32 Gain_hi_mean 2 Average gain vector Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2	Float32	dico2_isf_hf	1024	2nd stage isf codebook	
Float32 Sin20 20 Random phase Float32 Inter4_2 65 ¼ resolution interpolation filter Float32 VadFiltBandFreqs 12 Open-loop classifier Float32 Bw 12 Open-loop classifier Float32 Lwg 8 Open-loop classifier Float32 Gain_hf_ramp 64 HF gain ramp for wb->wb+ switching Float32 Inter2_coef 12 Filter coefficients for band join/split Float32 Filter_LP180 2341 Filter for 48 kHz interpolation Float32 StereoNbits 18 Stereo bit-rates Float32 Filter_2k 321 2k decimation filter Float32 Cb_filt_hi_mean 9 Average filter Float32 Filt_hi_mscb4a 16*9 Float32 Filt_hi_mscb_7a 16*9 Float32 Filt_hi_mscb_7b 8*9 Float32 Gain_hi_mean 2 Average gain vector Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2		Lag_window			
Float32 Inter4_2 65 % resolution interpolation filter Float32 VadFiltBandFreqs 12 Open-loop classifier Float32 Bw 12 Open-loop classifier Float32 Lwg 8 Open-loop classifier Float32 Gain_hf_ramp 64 HF gain ramp for wb->wb+ switching Float32 Inter2_coef 12 Filter coefficients for band join/split Float32 Filter_LP180 2341 Filter for 48 kHz interpolation Float32 StereoNbits 18 Stereo bit-rates Float32 Filter_2k 321 2k decimation filter Float32 Cb_filt_hi_mean 9 Average filter Float32 Filt_hi_mscb4a 16*9 Float32 Filt_hi_mscb_7a 16*9 Float32 Filt_hi_mscb_7b 8*9 Float32 Cb_gain_hi_mean 2 Average gain vector Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2	Float32	•			
Float32 VadFiltBandFreqs 12 Open-loop classifier Float32 Bw 12 Open-loop classifier Float32 Lwg 8 Open-loop classifier Float32 Gain_hf_ramp 64 HF gain ramp for wb->wb+ switching Float32 Inter2_coef 12 Filter coefficients for band join/split Float32 Filter_LP180 2341 Filter for 48 kHz interpolation Float32 StereoNbits 18 Stereo bit-rates Float32 Filter_2k 321 2k decimation filter Float32 Cb_filt_hi_mean 9 Average filter Float32 Filt_hi_mscb4a 16*9 Float32 Filt_hi_mscb4a 16*9 Float32 Filt_hi_mscb_7a 16*9 Float32 Filt_hi_mscb_7b 8*9 Float32 Cb_gain_hi_mean 2 Average gain vector Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2					
Float32 Bw 12 Open-loop classifier Float32 Lwg 8 Open-loop claissifier Float32 Gain_hf_ramp 64 HF gain ramp for wb->wb+ switching Float32 Inter2_coef 12 Filter coefficients for band join/split Float32 Filter_LP180 2341 Filter for 48 kHz interpolation Float32 StereoNbits 18 Stereo bit-rates Float32 Filter_2k 321 2k decimation filter Float32 Cb_filt_hi_mean 9 Average filter Float32 Filt_hi_mscb4a 16*9 Float32 Filt_hi_mscb4a 16*9 Float32 Filt_hi_mscb_7a 16*9 Float32 Filt_hi_mscb_7b 8*9 Float32 Cb_gain_hi_mean 2 Average gain vector Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2	Float32	Inter4_2		1/4 resolution interpolation filter	
Float32 Lwg 8 Open-loop claissifier Float32 Gain_hf_ramp 64 HF gain ramp for wb->wb+ switching Float32 Inter2_coef 12 Filter coefficients for band join/split Float32 Filter_LP180 2341 Filter for 48 kHz interpolation Float32 StereoNbits 18 Stereo bit-rates Float32 Filter_2k 321 2k decimation filter Float32 Cb_filt_hi_mean 9 Average filter Float32 Filt_hi_mscb4a 16*9 Float32 Filt_hi_mscb_7a 16*9 Float32 Filt_hi_mscb_7b 8*9 Float32 Cb_gain_hi_mean 2 Average gain vector Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2		VadFiltBandFreqs			
Float32		Bw			
Float32				·	
Float32 Filter_LP180 2341 Filter for 48 kHz interpolation Float32 StereoNbits 18 Stereo bit-rates Float32 Filter_2k 321 2k decimation filter Float32 Cb_filt_hi_mean 9 Average filter Float32 Filt_hi_mscb4a 16*9 Float32 Filt_hi_mscb_7a 16*9 Float32 Filt_hi_mscb_7b 8*9 Float32 Cb_gain_hi_mean 2 Average gain vector Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2					
Float32 StereoNbits 18 Stereo bit-rates Float32 Filter_2k 321 2k decimation filter Float32 Cb_filt_hi_mean 9 Average filter Float32 Filt_hi_mscb4a 16*9 Float32 Filt_hi_mscb_7a 16*9 Float32 Filt_hi_mscb_7b 8*9 Float32 Cb_gain_hi_mean 2 Average gain vector Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2					
Float32 Filter_2k 321 2k decimation filter Float32 Cb_filt_hi_mean 9 Average filter Float32 Filt_hi_mscb4a 16*9 Float32 Filt_hi_mscb_7a 16*9 Float32 Filt_hi_mscb_7b 8*9 Float32 Cb_gain_hi_mean 2 Average gain vector Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2					
Float32 Cb_filt_hi_mean 9 Average filter Float32 Filt_hi_mscb4a 16*9 Float32 Filt_hi_mscb_7a 16*9 Float32 Filt_hi_mscb_7b 8*9 Float32 Cb_gain_hi_mean 2 Average gain vector Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2					
Float32 Filt_hi_mscb4a 16*9 Float32 Filt_hi_mscb_7a 16*9 Float32 Filt_hi_mscb_7b 8*9 Float32 Cb_gain_hi_mean 2 Average gain vector Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2	Float32 Filter_2k 321 2k decim				
Float32 Filt_hi_mscb_7a 16*9 Float32 Filt_hi_mscb_7b 8*9 Float32 Cb_gain_hi_mean 2 Average gain vector Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2				Average filter	
Float32 Filt_hi_mscb_7b 8*9 Float32 Cb_gain_hi_mean 2 Average gain vector Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2					
Float32 Cb_gain_hi_mean 2 Average gain vector Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2					
Float32 Gain_hi_mscb_2a 4*2 Float32 Gain_hi_mscb_5a 32*2					
Float32 Gain_hi_mscb_5a 32*2				Average gain vector	
	Float32	Gain_hi_mscb_2a	4*2		
TBC	Float32	Gain_hi_mscb_5a	32*2		
		TBC			

Table 4: Decoder fixed tables

Format	Table name	Size	Description		
Same as encoder					

4.4.2 Static variables used in the C-code

In this clause two tables that specify the static variables for the encoder and decoder respectively are shown. All static variables are declared within a C **struct.**

Table 5: Encoder static variables

struct name	type	variable	size	description
Coder StState	71			•
	float	mem_decim	1608	speech decimated filter memory
	int	decim_frac	1	Fractional decimation factor
	float	mem_sig_in	4	hp filter memory
	float	mem_preemph	1	speech preemphasis filter mem
	float	mem_decim_hf	46	HF filter memory
	float	old_speech_hf	528	HF old speech vector
	float	past_q_isf_hf	8	HF past quantized isf
	float	ispold_hf	8	HF old isp
	float	ispold_q_hf	8	HF quantized old isp
	float	old_gain;	1	HF old gain match
	float	mem_hf1	8	HF memory for gain 1
	float	mem_hf2	8	HF memory for gain 2
	float	mem_hf3	8	HF memory for gain 3
	float	old_exc	375	old excitation
	float*	mean_isf_hf	1	isf codebook mean
	float*	dico1_isf_hf	1	isf codebook first stage
Coder_State_Plus				
	Coder_StState	left	2614	state for left channel
	Coder_StState	right	2614	state for right channel
	float	old_chan	528	old left signal
	float	old_chan_2k	140	old left signal 2kHz sampl. rate
	float	old_chan_hi	448	old left signal HB
			140	old mono signal 2kHz sampl.
	float	old_speech_2k	448	rate
	float	old_speech_hi	528	old mono signal HB
	float	old_speech_pe		past pre-emphasised mono
	float	old_wh	9	past weighted filter
	float	old_wh_q	2	past quantized weighted filter
	float	old_gm_gain	9	past gain matching
	float	old_exc_mono	+,	past mono excitation
	float	filt_energy_threshold	1 64	filter energy thershold
	float	w_window	1	weighting window
	PMSVQ*	*filt_hi_pmsvq	1	MSVQ quantizer
	PMSVQ*	*gain_hi_pmsvq	1	MSVQ quantizer
	int	mem_stereo_ovlp_size	32	past stereo overlap size
	float	mem_stereo_ovlp	1	past stereo overlap
	NCLASSDATA	*stClass	1	use case B classifier
	VadVars	*vadSt	1	VAD state
	short	vad_hist	528	VAD history
	float	old_speech	16	old speech
	float	old_synth	16	synthesis memory
	float	past_isfq	128	past isf quantizer
	float	old_wovlp	187	last tcx overlap
	float	old_d_wsp	392	Weighted speech vector
	float	old_exc	1	old excitation vector
	float	old_mem_wsyn	1	weighted synthesis memory
	float	old_mem_w0	1	weighted speech memory
	float	old_mem_xnq	I	quantized target memory

int	old_ovlp_size	1	last tcx overlap size
float	isfold	16	old isf frequency domain
float	ispold	16	old isp
float	ispold_q	16	quantized old isp
float	mem_wsp	1	wsp vector mem
float	mem_lp_decim2	3	wsp decimator filter mem
float	ada_w	1	open loop LTP
float	ol_gain	1	open loop LTP
short	ol_wght_flg	1	open loop LTP
long int	old_ol_lag	5	past openloop lag
int	old_T0_med	1	past pitch
float	hp_old_wsp	699	past HP weighted speech
float	hp_ol_ltp_mem	7	past HP openloop long term prediction
float	window	512	LP analysis window
short	SwitchFlagPlusToWB	1	flag for switching to AMR-WB
float	mem_gain_code	4	past code gain
short	prev_mod	1	past frame type

Table 6: Decoder static variables

struct name	type	variable	size	description
Decoder_StState				•
	float	mem_oversamp	72	Memory oversampling
	int	over_frac	1	Fractional overcloking factor
	float	mem_oversamp_hf	24	memory
	float	past_q_isf_hf	8	HF past quantized isf
			8	HF past quantized isf for the other
	float	past_q_isf_hf_other	1	channel when mono decoding stereo
	float	past_q_gain_hf	1	HF past quantized gain
	float	past_q_gain_hf_other		HF past quantized gain for the other channel when mono decoding stereo
	float	old_gain	1	HF old gain match
	float	ispold_hf	8	HF old isp
	float	threshold;	1	HF memory for smooth ener
	float	mem_syn_hf	8	HF synthesis memory
	float	mem_d_tcx	96	delay compensation memory
	float	mem_d_nonc	64	Non causality delay
	float	mem_synth_hi	16	High band sunthesis memory
	float	mem_sig_out	4	hp filter memory
	float	old_synth_hf	512	synch delay memory
	float	lp_amp	1	memory for soft exc
	float*	mean_isf_hf	1	isf codebook mean
	float*	dico1_isf_hf	1	isf codebook first stage
Decoder_State_Plus				
	Decoder_StState	left	828	State for left channel
	Decoder_StState	right	828	State for right channel
	float	mem_left_2k	20	2kHz memory on left chan
	float	mem_right_2k	20	2kHz memory on right chan
	float	mem_left_hi	64	HB memory left channel
	float	mem_right_hi	64	HB memory right channel
	float	my_old_synth_2k	35	old 2kHz synthesis
	float	my_old_synth_hi	128	old HB synthesis
	float	my_old_synth	148	old stereo synth
	float	old_AqLF	85	old quantized LPC
	float	old_wh	9	old decoded filter
	float	old_wh2	9	old decoded filter 2
	float	old_exc_mono	9	old mono excitation
	float	old_gain_left	4	old gain on left chan
	float	old_gain_right	4	old gain on right chan
	float	old_wh_q	9	past quantized filter
	float	old_gm_gain	2	past gain matching
	float	w_window	64	weighted synthesis window
	PMSVQ	*filt_hi_pmsvq	1	past MSVQ filter
	PMSVQ	*gain_hi_pmsvq	1	past MSVQ gain
	int	mem_stereo_ovlp_size	1	past stereo overlap size
	float	mem_stereo_ovlp	32	past stereo overlap
	int	last_stereo_mode	1	past stereo mode
	float	side_rms	1	side signal RMS
	float	h	9	current filter
	float	mem_balance	1	past balance factor

int		fer_hist	500	frame errasure history
int		fer_hist_ptr	1	frame erasure pointer
float	<u> </u>	fer_mean	1	frame erasure mean
float	<u> </u>	old_xri	1148	old spectral coefficeints
int		last_mode	1	last mode in previous 80ms frame
float	t	mem_sig_out	4	hp50 filter memory for synthesis
float	t	mem_deemph	1	speech deemph filter memory
int		prev_lpc_lost	1	previous lpc is lost when = 1
float	t	old_synth	16	synthesis memory
float	t	old_exc	392	old excitation vector
float	t	isfold	16	old isf (frequency domain)
float	t	ispold	16	old isp (immittance spectral pairs)
float	t	past_isfq	16	past isf quantizer
float	t	wovlp	128	last weighted synthesis for overlap
int		ovlp_size	1	overlap size
float	t	isf_buf	51	old isf (for frame recovery)
int		old_T0	1	old pitch value (for frame recovery)
int		old_T0_frac	1	old pitch value (for frame recovery)
shor	rt	seed_ace	1	seed memory (for random function)
float	t	mem_wsyn	1	TCX synthesis memory
shor	rt	seed_tcx	1	seed memory (for random function)
float	t	wsyn_rms	1	rms value of weighted synthesis
float	t .	past_gpit	1	past gain of pitch (for frame recovery)
float	t .	past_gcode	1	past gain of code (for frame recovery)
int		pitch_tcx	1	for bfi
float	t	gc_threshold	1	GC threshold
float	t	old_synth_pf	503	Bass post-filter: old synthesis
float	t	olu_noise_pi	24	bass post-filter: noise memory
int		old_1_pi	2	bass post-filter: old pitch
float	t	old_gain_pf	2	Bass post-filter: old pitch gain
float	<u>t</u>	*mean_isf_hf	1	HF isf codebook in-use
float	<u>t</u>	*dico1_isf_hf	1	HF isf codebook in-use
float	<u>t</u>	mem_gain_code	4	past code gain
float	t	mem_lpc_hf	9	past HF lpc filter
float	<u>t</u>	mem_gain_hf	1	past HF gain
shor	rt	ramp_state	1	ramp state

5 File formats

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

5.1 Audio file (encoder input/decoder output)

Audio files read by the encoder must be formatted as 16 bits PCM wave (*.wav) files. The decoder output is written as a 16 bit PCM wave file (*.wav).

Note that the decoder, with proper command line switch, can produce a mono file from a stereo bit-stream.

5.2 Parameter bitstream file (encoder output/decoder input)

For AMR-WB+ operation, the files produced by the audio encoder/expected by the audio decoder are either according to the raw format defined in Reference [2] Section 8.2, or according to the 3GP file format [4], whereby the storage sample definition is found in Reference [2] Section 8.3.

Annex A (informative): Change history

Change history							
Date	TSG SA#	TSG Doc.	CR	Rev	Subject/Comment	Old	New
2004-09	25	SP-040640	-	-	Approved at TSG SA#25	2.0.0	6.0.0
2004-12	26	SP-040841	001		ncorrect definition of mode index for SID frames		6.1.0
2004-12	26	SP-040841	002		Correction of TCX coding selection for MMS encoder	6.0.0	6.1.0
2004-12	26	SP-040841	003		Misread of energy buffer in coding mode selection in	6.0.0	6.1.0
					MMS encoder. Correction of energy buffer initialisation		
2004-12	26	SP-040841	004		Correction of stereo bit allocation tables	6.0.0	6.1.0
2004-12	26	SP-040841	005	1	Optimization of error concealment operation	6.0.0	6.1.0
2004-12	26	SP-040841	006	1	Stereo operation of pre-echo mode, saturation of gain_shape	6.0.0	6.1.0
2004-12	26	SP-040841	007		Stereo operation of pre-echo mode, alignment of encoder and decoder	6.0.0	6.1.0
2004-12	26	SP-040841	008	1	Addition of support for file formats and improved	6.0.0	6.1.0
					command line		
2004-12	26	SP-040841	009	1	Source code editorial changes	6.0.0	6.1.0
2004-12	26	SP-040841	010		Removal of complexity counters	6.0.0	6.1.0
2004-12	26	SP-040841	011	1	Editorial changes	6.0.0	6.1.0
2004-12	26	SP-040841	012		Void. This CR in S4-040722 (Title: Editorial changes) was meant to TS 26.290 (as CR 004) and not to TS 26.304! The CR was implemented instead in TS 26.290 v. 6.1.0 and a remark was put in the CR database.	6.0.0	6.1.0
2004-12	26	SP-040841	013		Removal of the eid tool	6.0.0	6.1.0
2004-12	26	SP-040841	014	1	Addition of frame erasure simulation at the decoder	6.0.0	6.1.0
2004-12	26	SP-040841	015		Removal of two unused stereo rates	6.0.0	6.1.0
2004-12	26	SP-040841	016		Removal of extStMode	6.0.0	6.1.0
2005-03	27	SP-050096	019	1	AMR-WB/AMR-WB+ switching	6.1.0	6.2.0
2005-03	27	SP-050096	020	2	Clean-up of unused C-code functions	6.1.0	6.2.0
2005-03	27	SP-050096	021	1	Correction of misbehaviour of constrained cholesky	6.1.0	6.2.0
2005-03	27	SP-050096	022	1	Source code bit exact editorial changes	6.1.0	6.2.0
2005-03	27	SP-050096	023	2	Correction of last frame processing	6.1.0	6.2.0
2005-03	27	SP-050096	024	1	Correction of frame erasure concealment	6.1.0	6.2.0
2005-03	27	SP-050096	025	1	Correction of references and terminology	6.1.0	6.2.0
2005-06	28	SP-050252	026		Correction of DTX handling in AMR-WB modes	6.2.0	6.3.0
2005-06	28	SP-050252	027		Remove IF2 header in AMR-WB bitstream	6.2.0	6.3.0
2005-06	28	SP-050252	028		Decoder synchronization after frame erasures	6.2.0	6.3.0
2005-06	28	SP-050252	029		Correction for buffer reading in low complexity encoder	6.2.0	6.3.0
2005-06	28	SP-050252	030		Correction to a wrong function call	6.2.0	6.3.0
2005-06	28	SP-050252	031		Correction of mode switching using configuration file	6.2.0	6.3.0
2005-06	28	SP-050252	032		Correction of information printed by decoder in DTX frames	6.2.0	6.3.0
2005-06	28	SP-050252	033		Correction of library function for 3GP file format	6.2.0	6.3.0
2005-06	28	SP-050252			Support for input files with sampling frequency other than 48 kHz	6.2.0	6.3.0
2005-09	29	SP-050425	0035		Correction to frame erasure concealment	6.3.0	6.4.0
2006-03	31	SP-060012			Correction to end-of-file logic and initialisation in AMR-WB modes	6.4.0	6.5.0
2006-03	31	SP-060012	0037		Correction to unnecessary look ahead in encoder	6.4.0	6.5.0
2006-03	31	SP-060012			Correction to differencessary look aread in encoder Correction to memory initialization and memory overwrite when switching between AMR-WB and AMR-	6.4.0	6.5.0
					WB+ modes		
2006-03	31	SP-060012	0039		Correction to VC 6.0 compilation warnings	6.4.0	6.5.0

History

	Document history					
V6.1.0	December 2004	Publication				
V6.2.0	March 2005	Publication				
V6.3.0	June 2005	Publication				
V6.4.0	September 2005	Publication				
V6.5.0	March 2006	Publication				