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3rd Generation Partnership Project;

Technical Specification Group Services and System Aspects;

Speech codec speech processing functions;

Adaptive Multi-Rate - Wideband (AMR-WB) speech codec;

ANSI-C code

(Release 16)

** 

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# Foreword

This Technical Specification (TS) 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:

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where:

x the first digit:

1 presented to TSG for information;

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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 Adaptive Multi-Rate Wideband codec. This floating-point codec specification is mainly targeted to be used in multimedia applications or in packet-based applications. The bit-exact fixed-point ANSI-C code in 3GPP TS 26.173 remains the preferred implementation for all applications, but the floating-point codec may be used instead of the fixed-point codec when the implementation platform is better suited for a floating-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 is the only standard conforming non-bit-exact implementation of the Adaptive Multi-Rate Wideband speech transcoder (3GPP TS 26.190 [2]), Voice Activity Detection (3GPP TS 26.194 [6]), comfort noise generation (3GPP TS 26.192 [4]), and source controlled rate operation (3GPP TS 26.193 [5]). The floating-point code also contains example solutions for substituting and muting of lost frames (3GPP TS 26.191 [3]).

The fixed-point specification in 26.173 shall remain the only allowed implementation for the 3G AMR-WB speech service and the use of the floating-point codec is strictly limited to other services.

The floating-point encoder in the present document is a non-bit-exact implementation of the fixed-point encoder producing quality indistinguishable from that of the fixed-point encoder. The decoder in the present document is functionally a bit‑exact implementation of the fixed-point decoder, but the code has been optimized for speed and the standard fixed-point libraries are not used as such.

# 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 26.174: "AMR speech codec, wideband; Test sequences".

[2] 3GPP TS 26.190: "Mandatory Speech Codec speech processing functions AMR Wideband speech codec; Transcoding functions".

[3] 3GPP TS 26.191: "AMR speech codec, wideband; Error concealment of lost frames".

[4] 3GPP TS 26.192: "Mandatory Speech Codec speech processing functions AMR Wideband Speech Codec; Comfort noise aspects".

[5] 3GPP TS 26.193: "AMR speech codec, wideband; Source controlled rate operation".

[6] 3GPP TS 26.194: "Mandatory Speech Codec speech processing functions AMR Wideband speech codec; Voice Activity Detector (VAD)".

[7] RFC 3267 "A Real-Time Transport Protocol (RTP) Payload Format and File Storage Format for Adaptive Multi-Rate (AMR) and Adaptive Multi-Rate Wideband (AMR-WB) Audio Codecs", June 2002.

# 3 Definitions and abbreviations

## 3.1 Definitions

For the purposes of the present document, the terms and definitions given in TS 26.190 [2], TS 26.191 [3], TS 26.192 [4], TS 26.193 [5] and TS 26.194 [6].

## 3.2 Abbreviations

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

AMR-WB 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 bit‑exact 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 NT40 and Microsoft Visual C++ v.6.0 compiler.

- IBM PC/AT compatible computers with Windows NT40 and Intel C/C++ v.4.0 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 all files in the root level.

The distributed files with suffix "c" contain the source code and the files with suffix "h" are the header files. The ROM data is contained in "rom" files with suffix "c".

Makefiles are provided for the platforms in which the C code has been verified (listed above). Once the software is installed, this directory will have a compiled version of encoder and decoder and all the object files.

## 4.2 Program execution

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

*-* (*encoder*) speech encoder;

*-* (*decoder*) speech decoder.

The programs should be called like:

- encoder [encoder options] <speech input file> <parameter file>;

- decoder <parameter file> <speech output file>.

The speech files contain 16-bit linear encoded PCM speech samples and the parameter files contain encoded speech 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 speech codec, including the functions of VAD, DTX, and comfort noise generation.

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: Speech encoder call structure

|  |  |  |  |  |
| --- | --- | --- | --- | --- |
| E\_MAIN\_encode | E\_UTIL\_decim\_12k8 | E\_UTIL\_down\_samp | E\_UTIL\_interpol |  |
|  | E\_UTIL\_decim\_12k8 |  |  |  |
|  | E\_UTIL\_hp50\_12k8 |  |  |  |
|  | E\_UTIL\_hp50\_12k8 |  |  |  |
|  | E\_UTIL\_f\_preemph |  |  |  |
|  | E\_DTX\_vad | E\_DTX\_filter\_bank | E\_DTX\_filter5 |  |
|  |  |  | E\_DTX\_filter3 |  |
|  |  |  | E\_DTX\_level\_calculation |  |
|  |  | E\_DTX\_decision | E\_DTX\_noise\_estimate\_update | E\_DTX\_update\_cntrl |
|  |  |  | E\_DTX\_hangover\_addition |  |
|  |  | E\_DTX\_speech\_estimate |  |  |
|  | E\_DTX\_tx\_handler |  |  |  |
|  | E\_DTX\_reset | E\_LPC\_isf\_init |  |  |
|  | E\_MAIN\_parm\_store |  |  |  |
|  | E\_UTIL\_autocorr |  |  |  |
|  | E\_LPC\_lag\_wind |  |  |  |
|  | E\_LPC\_lev\_dur |  |  |  |
|  | E\_LPC\_a\_isp\_conversion | E\_LPC\_chebyshev |  |  |
|  | E\_LPC\_f\_int\_isp\_find | E\_LPC\_f\_isp\_a\_conversion | E\_LPC\_f\_isp\_pol\_get |  |
|  | E\_LPC\_isp\_isf\_conversion |  |  |  |
|  | E\_GAIN\_clip\_isf\_test |  |  |  |
|  | E\_LPC\_a\_weight |  |  |  |
|  | E\_UTIL\_residu |  |  |  |
|  | E\_UTIL\_deemph |  |  |  |
|  | E\_GAIN\_lp\_decim2 |  |  |  |
|  | E\_GAIN\_open\_loop\_search |  |  |  |
|  | E\_GAIN\_olag\_median | E\_GAIN\_sort |  |  |
|  | E\_DTX\_pitch\_tone\_detection |  |  |  |
|  | E\_GAIN\_open\_loop\_search |  |  |  |
|  | E\_GAIN\_olag\_median |  |  |  |
|  | E\_DTX\_pitch\_tone\_detection |  |  |  |
|  | E\_UTIL\_residu |  |  |  |
|  | E\_DTX\_buffer |  |  |  |
|  | E\_DTX\_exe | E\_DTX\_frame\_indices\_find |  |  |
|  |  | E\_DTX\_isf\_history\_aver |  |  |
|  |  | E\_DTX\_isf\_q | E\_LPC\_isf\_sub\_vq |  |
|  |  |  | E\_LPC\_isf\_noise\_d | E\_LPC\_f\_isf\_reorder |
|  |  | E\_DTX\_dithering\_control |  |  |
|  |  | E\_UTIL\_random |  |  |
|  | E\_MAIN\_reset | E\_GAIN\_clip\_init |  |  |
|  |  | E\_DTX\_reset |  |  |
|  |  | E\_DTX\_vad\_reset |  |  |
|  | E\_LPC\_isf\_2s3s\_quantise | E\_LPC\_stage1\_isf\_vq |  |  |
|  |  | E\_LPC\_isf\_sub\_vq |  |  |
|  |  | E\_LPC\_stage1\_isf\_vq |  |  |
|  |  | E\_LPC\_isf\_sub\_vq |  |  |
|  |  | E\_LPC\_isf\_2s3s\_decode | E\_LPC\_isf\_reorder |  |
|  | E\_LPC\_isf\_2s5s\_quantise | E\_LPC\_stage1\_isf\_vq |  |  |
|  |  | E\_LPC\_isf\_sub\_vq |  |  |
|  |  | E\_LPC\_isf\_2s5s\_decode | E\_LPC\_isf\_reorder |  |
|  | E\_LPC\_isf\_isp\_conversion |  |  |  |
|  | E\_LPC\_int\_isp\_find | E\_LPC\_isp\_a\_conversion | E\_LPC\_isp\_pol\_get | E\_UTIL\_l\_extract |
|  |  |  |  | E\_UTIL\_mpy\_32\_16 |
|  |  |  | E\_UTIL\_l\_extract |  |
|  |  |  | E\_UTIL\_mpy\_32\_16 |  |
|  | E\_UTIL\_residu |  |  |  |
|  | E\_DTX\_buffer |  |  |  |
|  | E\_UTIL\_residu |  |  |  |
|  | E\_UTIL\_synthesis |  |  |  |
|  | E\_LPC\_a\_weight |  |  |  |
|  | E\_UTIL\_residu |  |  |  |
|  | E\_UTIL\_deemph |  |  |  |
|  | E\_UTIL\_f\_preemph |  |  |  |
|  | E\_LPC\_a\_weight |  |  |  |
|  | E\_UTIL\_synthesis |  |  |  |
|  | E\_UTIL\_residu |  |  |  |
|  | E\_LPC\_a\_weight |  |  |  |
|  | E\_UTIL\_synthesis |  |  |  |
|  | E\_UTIL\_deemph |  |  |  |
|  | E\_GAIN\_closed\_loop\_search | E\_GAIN\_norm\_corr | E\_UTIL\_f\_convolve |  |
|  |  | E\_GAIN\_norm\_corr\_interpolate |  |  |
|  | E\_GAIN\_clip\_test |  |  |  |
|  | E\_GAIN\_adaptive\_codebook\_excitation |  |  |  |
|  | E\_UTIL\_convolve |  |  |  |
|  | E\_ACELP\_xy1\_corr |  |  |  |
|  | E\_ACELP\_codebook\_target\_update |  |  |  |
|  | E\_UTIL\_convolve |  |  |  |
|  | E\_ACELP\_xy1\_corr |  |  |  |
|  | E\_ACELP\_codebook\_target\_update |  |  |  |
|  | E\_UTIL\_f\_preemph |  |  |  |
|  | E\_GAIN\_f\_pitch\_sharpening |  |  |  |
|  | E\_ACELP\_xh\_corr |  |  |  |
|  | E\_ACELP\_2t |  |  |  |
|  | E\_ACELP\_4t | E\_ACELP\_h\_vec\_corr1 |  |  |
|  |  | E\_ACELP\_h\_vec\_corr2 |  |  |
|  |  | E\_ACELP\_2pulse\_search |  |  |
|  |  | E\_ACELP\_quant\_1p\_N1 |  |  |
|  |  | E\_ACELP\_quant\_2p\_2N1 |  |  |
|  |  | E\_ACELP\_quant\_3p\_3N1 | E\_ACELP\_quant\_2p\_2N1 |  |
|  |  |  | E\_ACELP\_quant\_1p\_N1 |  |
|  |  | E\_ACELP\_quant\_4p\_4N | E\_ACELP\_quant\_4p\_4N1 | E\_ACELP\_quant\_2p\_2N1 |
|  |  |  | E\_ACELP\_quant\_1p\_N1 |  |
|  |  |  | E\_ACELP\_quant\_3p\_3N1 |  |
|  |  |  | E\_ACELP\_quant\_2p\_2N1 |  |
|  |  |  | E\_ACELP\_quant\_3p\_3N1 |  |
|  |  | E\_ACELP\_quant\_5p\_5N | E\_ACELP\_quant\_3p\_3N1 |  |
|  |  |  | E\_ACELP\_quant\_2p\_2N1 |  |
|  |  | E\_ACELP\_quant\_6p\_6N\_2 | E\_ACELP\_quant\_5p\_5N |  |
|  |  |  | E\_ACELP\_quant\_1p\_N1 |  |
|  |  |  | E\_ACELP\_quant\_4p\_4N |  |
|  |  |  | E\_ACELP\_quant\_2p\_2N1 |  |
|  |  |  | E\_ACELP\_quant\_3p\_3N1 |  |
|  | E\_UTIL\_preemph |  |  |  |
|  | E\_GAIN\_pitch\_sharpening |  |  |  |
|  | E\_ACELP\_xy2\_corr |  |  |  |
|  | E\_ACELP\_gains\_quantise | E\_UTIL\_dot\_product12 | E\_UTIL\_saturate\_31 |  |
|  |  |  | E\_UTIL\_norm\_l |  |
|  |  | E\_UTIL\_normalized\_inverse\_sqrt |  |  |
|  |  | E\_UTIL\_l\_extract |  |  |
|  |  | E\_UTIL\_saturate |  |  |
|  |  | E\_UTIL\_mpy\_32\_16 |  |  |
|  |  | E\_UTIL\_log2\_32 | E\_UTIL\_norm\_l |  |
|  |  |  | E\_UTIL\_normalized\_log2 |  |
|  | E\_UTIL\_signal\_up\_scale |  |  |  |
|  | E\_UTIL\_signal\_down\_scale |  |  |  |
|  | E\_GAIN\_clip\_pit\_test |  |  |  |
|  | E\_UTIL\_signal\_down\_scale |  |  |  |
|  | E\_GAIN\_voice\_factor | E\_UTIL\_dot\_product12 |  |  |
|  |  | E\_UTIL\_norm\_l |  |  |
|  |  | E\_UTIL\_norm\_s |  |  |
|  | E\_UTIL\_norm\_s |  |  |  |
|  | E\_UTIL\_synthesis |  |  |  |
|  | E\_UTIL\_enc\_synthesis | E\_UTIL\_synthesis |  |  |
|  |  | E\_UTIL\_deemph |  |  |
|  |  | E\_UTIL\_hp50\_12k8 |  |  |
|  |  | E\_UTIL\_random |  |  |
|  |  | E\_UTIL\_hp400\_12k8 |  |  |
|  |  | E\_LPC\_a\_weight |  |  |
|  |  | E\_UTIL\_synthesis |  |  |
|  |  | E\_UTIL\_bp\_6k\_7k |  |  |
|  |  | E\_UTIL\_bp\_6k\_7k |  |  |

Table 2: Speech decoder call structure

|  |  |  |  |  |
| --- | --- | --- | --- | --- |
| D\_MAIN\_decode | D\_DTX\_rx\_handler | D\_LPC\_isf\_noise\_d | D\_LPC\_isf\_reorder |  |
|  | D\_DTX\_exe | D\_DTX\_cn\_dithering | D\_UTIL\_random |  |
|  |  | D\_UTIL\_pow2 |  |  |
|  |  | D\_UTIL\_norm\_l |  |  |
|  |  | D\_UTIL\_random |  |  |
|  |  | D\_UTIL\_dot\_product12 | D\_UTIL\_norm\_l |  |
|  |  | D\_UTIL\_normalized\_inverse\_sqrt |  |  |
|  | D\_LPC\_isf\_isp\_conversion |  |  |  |
|  | D\_LPC\_isp\_a\_conversion | D\_LPC\_isp\_pol\_get | D\_UTIL\_l\_extract |  |
|  |  |  | D\_UTIL\_mpy\_32\_16 |  |
|  |  | D\_UTIL\_l\_extract |  |  |
|  |  | D\_UTIL\_mpy\_32\_16 |  |  |
|  | D\_UTIL\_dec\_synthesis | D\_UTIL\_synthesis\_32 |  |  |
|  |  | D\_UTIL\_deemph\_32 | D\_UTIL\_saturate |  |
|  |  | D\_UTIL\_hp50\_12k8 | D\_UTIL\_l\_extract |  |
|  |  | D\_UTIL\_oversamp\_16k | D\_UTIL\_up\_samp | D\_UTIL\_interpol |
|  |  | D\_UTIL\_random |  |  |
|  |  | D\_UTIL\_signal\_down\_scale |  |  |
|  |  | D\_UTIL\_dot\_product12 |  |  |
|  |  | D\_UTIL\_normalized\_inverse\_sqrt |  |  |
|  |  | D\_UTIL\_hp400\_12k8 | D\_UTIL\_l\_extract |  |
|  |  | D\_UTIL\_norm\_l |  |  |
|  |  | D\_LPC\_isf\_extrapolation | D\_UTIL\_norm\_s |  |
|  |  |  | D\_UTIL\_l\_extract |  |
|  |  |  | D\_UTIL\_mpy\_32 |  |
|  |  |  | D\_LPC\_isf\_isp\_conversion |  |
|  |  | D\_LPC\_isp\_a\_conversion |  |  |
|  |  | D\_LPC\_a\_weight |  |  |
|  |  | D\_UTIL\_synthesis |  |  |
|  |  | D\_LPC\_a\_weight |  |  |
|  |  | D\_UTIL\_synthesis |  |  |
|  |  | D\_UTIL\_bp\_6k\_7k |  |  |
|  |  | D\_UTIL\_hp\_7k |  |  |
|  | D\_MAIN\_reset | D\_GAIN\_init |  |  |
|  |  | D\_GAIN\_lag\_concealment\_init |  |  |
|  |  | D\_DTX\_reset |  |  |
|  | D\_LPC\_isf\_2s3s\_decode | D\_LPC\_isf\_reorder |  |  |
|  | D\_LPC\_isf\_2s5s\_decode | D\_LPC\_isf\_reorder |  |  |
|  | D\_LPC\_isf\_isp\_conversion |  |  |  |
|  | D\_LPC\_int\_isp\_find | D\_LPC\_isp\_a\_conversion |  |  |
|  | D\_GAIN\_lag\_concealment | D\_GAIN\_sort\_lag | D\_GAIN\_insert\_lag |  |
|  |  | D\_UTIL\_random |  |  |
|  | D\_GAIN\_adaptive\_codebook\_excitation |  |  |  |
|  | D\_UTIL\_random |  |  |  |
|  | D\_ACELP\_decode\_2t |  |  |  |
|  | D\_ACELP\_decode\_4t | D\_ACELP\_decode\_1p\_N1 |  |  |
|  |  | D\_ACELP\_add\_pulse |  |  |
|  |  | D\_ACELP\_decode\_2p\_2N1 |  |  |
|  |  | D\_ACELP\_decode\_3p\_3N1 | D\_ACELP\_decode\_2p\_2N1 |  |
|  |  |  | D\_ACELP\_decode\_1p\_N1 |  |
|  |  | D\_ACELP\_decode\_4p\_4N | D\_ACELP\_decode\_4p\_4N1 | D\_ACELP\_decode\_2p\_2N1 |
|  |  |  |  | D\_ACELP\_decode\_2p\_2N1 |
|  |  |  | D\_ACELP\_decode\_1p\_N1 |  |
|  |  |  | D\_ACELP\_decode\_3p\_3N1 |  |
|  |  |  | D\_ACELP\_decode\_2p\_2N1 |  |
|  |  | D\_ACELP\_decode\_5p\_5N | D\_ACELP\_decode\_3p\_3N1 |  |
|  |  |  | D\_ACELP\_decode\_2p\_2N1 |  |
|  |  | D\_ACELP\_decode\_6p\_6N\_2 | D\_ACELP\_decode\_5p\_5N |  |
|  |  |  | D\_ACELP\_decode\_1p\_N1 |  |
|  |  |  | D\_ACELP\_decode\_4p\_4N |  |
|  |  |  | D\_ACELP\_decode\_2p\_2N1 |  |
|  |  |  | D\_ACELP\_decode\_3p\_3N1 |  |
|  | D\_UTIL\_preemph |  |  |  |
|  | D\_GAIN\_pitch\_sharpening |  |  |  |
|  | D\_GAIN\_decode | D\_UTIL\_dot\_product12 |  |  |
|  |  | D\_UTIL\_normalized\_inverse\_sqrt |  |  |
|  |  | D\_GAIN\_median |  |  |
|  |  | D\_UTIL\_l\_extract |  |  |
|  |  | D\_UTIL\_pow2 |  |  |
|  |  | D\_UTIL\_mpy\_32\_16 |  |  |
|  |  | D\_UTIL\_log2 | D\_UTIL\_norm\_l |  |
|  |  |  | D\_UTIL\_normalized\_log2 |  |
|  | D\_UTIL\_signal\_up\_scale | D\_UTIL\_saturate |  |  |
|  | D\_UTIL\_signal\_down\_scale |  |  |  |
|  | D\_GAIN\_find\_voice\_factor | D\_UTIL\_dot\_product12 |  |  |
|  |  | D\_UTIL\_norm\_l |  |  |
|  |  | D\_UTIL\_norm\_s |  |  |
|  | D\_UTIL\_norm\_s |  |  |  |
|  | D\_UTIL\_l\_extract |  |  |  |
|  | D\_ACELP\_phase\_disper |  |  |  |
|  | D\_UTIL\_mpy\_32\_16 |  |  |  |
|  | D\_UTIL\_l\_extract |  |  |  |
|  | D\_GAIN\_adaptive\_control | D\_UTIL\_norm\_l |  |  |
|  |  | D\_UTIL\_inverse\_sqrt |  |  |
|  | D\_UTIL\_dec\_synthesis | D\_UTIL\_saturate |  |  |
|  | D\_UTIL\_signal\_down\_scale |  |  |  |
|  | D\_DTX\_activity\_update | D\_UTIL\_log2 |  |  |

## 4.4 Variables, constants and tables

The data types of variables and tables used in the floating-point implementation are signed integers in 2's complement representation, defined by:

**Word8** 8 bit variable **UWord8** 8 bit unsigned variable

**Word16** 16 bit variable  
**Word16** 16 bit unsigned variable  
**Word32** 32 bit variable

Floating-point numbers use the IEEE (Institute of Electrical and Electronics Engineers) format:

**Float32** 8 bit exponent, 23 bit mantissa, 1 bit sign   
**Float64** 11 bit exponent, 52 bit mantissa, 1 bit sign

### 4.4.1 Description of fixed tables used in the C-code

This clause contains a listing of all fixed tables declared in enc\_rom.c and dec\_rom.c files.

Table 3: Encoder fixed tables

| **Format** | **Table name** | **Size** | **Description** |
| --- | --- | --- | --- |
| Word16 | E\_ROM\_cdown\_unusable | 7 | Attenuation factors for codebook gain in lost frames |
| Word16 | E\_ROM\_cdown\_usable | 7 | Attenuation factors for codebook gain in bad frames |
| Float32. | E\_ROM\_corrweight | 199 | Weighting of the correlation function in open loop LTP search |
| Word16 | E\_ROM\_cos | 129 | Table of cos(x) |
| Float32 | E\_ROM\_dico1\_isf | 9\*256 | 1st ISF quantizer of the 1st stage |
| Float32 | E\_ROM\_dico1\_isf\_noise | 2\*64 | 1st ISF quantizer for comfort noise |
| Float32 | E\_ROM\_dico21\_isf | 3\*64 | 1st ISF quantizer of the 2nd stage (not the 6.60 kbit/s mode) |
| Float32 | E\_ROM\_dico21\_isf\_36b | 5\*128 | 1st ISF quantizer of the 2nd stage (the 6.60 kbit/s mode) |
| Float32 | E\_ROM\_dico22\_isf | 3\*128 | 2nd ISF quantizer of the 2nd stage (not the 6.60 kbit/s mode) |
| Float32 | E\_ROM\_dico22\_isf\_36b | 4\*128 | 2nd ISF quantizer of the 2nd stage (the 6.60 kbit/s mode) |
| Float32 | E\_ROM\_dico23\_isf | 3\*128 | 3rd ISF quantizer of the 2nd stage (not the 6.60 kbit/s mode) |
| Float32 | E\_ROM\_dico23\_isf\_36b | 7\*64 | 3rd ISF quantizer of the 2nd stage (the 6.60 kbit/s mode) |
| Float32 | E\_ROM\_dico24\_isf | 3\*32 | 4th ISF quantizer of the 2nd stage (not the 6.60 kbit/s mode) |
| Float32 | E\_ROM\_dico25\_isf | 4\*32 | 5th ISF quantizer of the 2nd stage (not the 6.60 kbit/s mode) |
| Float32 | E\_ROM\_dico2\_isf | 7\*256 | 2nd ISF quantizer of the 1st stage |
| Float32 | E\_ROM\_dico2\_isf\_noise | 3\*64 | 2nd ISF quantizer for comfort noise |
| Float32 | E\_ROM\_dico3\_isf\_noise | 3\*64 | 3rd LSF quantizer for comfort noise |
| Float32 | E\_ROM\_dico4\_isf\_noise | 4\*32 | 4th LSF quantizer for comfort noise |
| Float32 | E\_ROM\_dico5\_isf\_noise | 4\*32 | 5th LSF quantizer for comfort noise |
| Float32 | E\_ROM\_en\_adjust | 9 | Energy scaling factor for each mode during comfort noise |
| Float32 | E\_ROM\_f\_interpol\_frac | 4 | LPC interpolation coefficients |
| Float32 | E\_ROM\_fir\_6k\_7k | 31 | Bandpass FIR filter coefficients for higher band generation |
| Word16 | E\_ROM\_fir\_down | 120 | Downsample FIR filter coefficients |
| Float32 | E\_ROM\_fir\_ipol | 61 | Interpol FIR filter coefficients |
| Word16 | E\_ROM\_fir\_up | 120 | Upsample FIR filter coefficients |
| Float32 | E\_ROM\_grid | 101 | Chebyshev polynomial grid points |
| Float32 | E\_ROM\_hamming\_cos | 384 | LP analysis window |
| Float32 | E\_ROM\_hp\_gain | 16 | High band gain table for 23.85 kbit/s mode |
| Float32 | E\_ROM\_inter4\_1 | 4\*2\*4 | Interpolation filter coefficients |
| Word16 | E\_ROM\_inter4\_2 | 4\*2\*16 | Interpolation filter coefficients |
| Word16 | E\_ROM\_interpol\_frac | 4 | Interpolation filter coefficients |
| Float32 | E\_ROM\_isf | 16 | ISF table for initialization |
| Word16 | E\_ROM\_isp | 16 | ISP table for initialization |
| Word16 | E\_ROM\_isqrt | 49 | Table used in inverse square root computation |
| Float32 | E\_ROM\_lag\_window | 16 | Lag window table |
| Word16 | E\_ROM\_log2 | 33 | Table used in logarithm computation |
| Float32 | E\_ROM\_f\_mean\_isf | 16 | ISF mean |
| Word16 | E\_ROM\_mean\_isf | 16 | ISF mean |
| Float32 | E\_ROM\_mean\_isf\_noise | 16 | ISF mean for comfort noise |
| Word16 | E\_ROM\_pdown\_unusable | 7 | Attenuation factors for adaptive codebook gain in lost frames |
| Word16 | E\_ROM\_pdown\_usable | 7 | Attenuation factors for adaptive codebook gain in bad frames |
| Word16 | E\_ROM\_pow2 | 33 | Table used in power of two computation |
| Float32 | E\_ROM\_qua\_gain6b | 2\*64 | Gain quantization table for 6-bit gain quantization |
| Float32 | E\_ROM\_qua\_gain7b | 2\*128 | Gain quantization table for 7-bit gain quantization |
| Uword8 | E\_ROM\_tipos | 36 | Starting point for codebook search |

Table 4: Decoder fixed tables

| **Format** | **Table name** | **Size** | **Description** |
| --- | --- | --- | --- |
| Word16 | D\_ROM\_cdown\_unusable | 7 | Attenuation factors for codebook gain in lost frames |
| Word16 | D\_ROM\_cdown\_usable | 7 | Attenuation factors for codebook gain in bad frames |
| Word16 | D\_ROM\_cos | 129 | Table of cos(x) |
| Word16 | D\_ROM\_dico1\_isf | 9\*256 | 1st ISF quantizer of the 1st stage |
| Word16 | D\_ROM\_dico1\_isf\_noise | 2\*64 | 1st ISF quantizer for comfort noise |
| Word16 | D\_ROM\_dico21\_isf | 3\*64 | 1st ISF quantizer of the 2nd stage (not the 6.60 kbit/s mode) |
| Word16 | D\_ROM\_dico21\_isf\_36b | 5\*128 | 1st ISF quantizer of the 2nd stage (the 6.60 kbit/s mode) |
| Word16 | D\_ROM\_dico22\_isf | 3\*128 | 2nd ISF quantizer of the 2nd stage (not the 6.60 kbit/s mode) |
| Word16 | D\_ROM\_dico22\_isf\_36b | 4\*128 | 2nd ISF quantizer of the 2nd stage (the 6.60 kbit/s mode) |
| Word16 | D\_ROM\_dico23\_isf | 3\*128 | 3rd ISF quantizer of the 2nd stage (not the 6.60 kbit/s mode) |
| Word16 | D\_ROM\_dico23\_isf\_36b | 7\*64 | 3rd ISF quantizer of the 2nd stage (the 6.60 kbit/s mode) |
| Word16 | D\_ROM\_dico24\_isf | 3\*32 | 4th ISF quantizer of the 2nd stage (not the 6.60 kbit/s mode) |
| Word16 | D\_ROM\_dico25\_isf | 5\*32 | 5th ISF quantizer of the 2nd stage (not the 6.60 kbit/s mode) |
| Word16 | D\_ROM\_dico2\_isf | 7\*256 | 2nd ISF quantizer of the 1st stage |
| Word16 | D\_ROM\_dico2\_isf\_noise | 3\*64 | 2nd ISF quantizer for comfort noise |
| Word16 | D\_ROM\_dico3\_isf\_noise | 3\*64 | 3rd LSF quantizer for comfort noise |
| Word16 | D\_ROM\_dico4\_isf\_noise | 4\*32 | 4th LSF quantizer for comfort noise |
| Word16 | D\_ROM\_dico5\_isf\_noise | 4\*32 | 5th LSF quantizer for comfort noise |
| Word16 | D\_ROM\_fir\_6k\_7k | 31 | Bandpass FIR filter coefficients for higher band generation |
| Word16 | D\_ROM\_fir\_7k | 31 | Bandpass FIR filter coefficients for higher band in 23.85 kbit/s mode |
| Word16 | D\_ROM\_fir\_down | 120 | Downsample FIR filter coefficients |
| Word16 | D\_ROM\_fir\_up | 120 | Upsample FIR filter coefficients |
| Word16 | D\_ROM\_hp\_gain | 16 | High band gain table for 23.85 kbit/s mode |
| Word16 | D\_ROM\_inter4\_2 | 4\*2\*16 | Interpolation filter coefficients |
| Word16 | D\_ROM\_interpol\_frac | 4 | LPC interpolation coefficients |
| Word16 | D\_ROM\_isf | 16 | ISF table for initialization |
| Word16 | D\_ROM\_isp | 16 | ISP table for initialization |
| Word16 | D\_ROM\_isqrt | 49 | Table used in inverse square root computation |
| Word16 | D\_ROM\_log2 | 33 | Table used in logarithm computation |
| Word16 | D\_ROM\_mean\_isf | 16 | ISF mean |
| Word16 | D\_ROM\_mean\_isf\_noise | 16 | ISF mean for comfort noise |
| Word16 | D\_ROM\_pdown\_unusable | 7 | Attenuation factors for adaptive codebook gain in lost frames |
| Word16 | D\_ROM\_pdown\_usable | 7 | Attenuation factors for adaptive codebook gain in bad frames |
| Word16 | D\_ROM\_ph\_imp\_low | 64 | Phase dispersion impulse response |
| Word16 | D\_ROM\_ph\_imp\_mid | 64 | Phase dispersion impulse response |
| Word16 | D\_ROM\_pow2 | 33 | Table used in power of two computation |
| Word16 | D\_ROM\_qua\_gain6b | 2\*64 | Gain quantization table for 6-bit gain quantization |
| Word16 | D\_ROM\_qua\_gain7b | 2\*128 | Gain quantization table for 7-bit gain quantization |

### 4.4.2 Static variables used in the C-code

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

Table 5: Speech encoder static variables

| **Struct name** | **Variable** | **Type** | **Length** | **Description** |
| --- | --- | --- | --- | --- |
| Coder\_State | mem\_speech | Float32 | 384 | speech buffer |
|  | mem\_wsp | Float32 | 371 | buffer holding spectral weighted speech |
|  | mem\_hp\_wsp | Float32 | 243 | highpass wsp |
|  | mem\_decim | Float32 | 30 | Open-loop LTP decimation filter memory |
|  | mem\_hf | Float32 | 30 | Estimated BP filter memory (23.85 kbit/s mode) |
|  | mem\_hf2 | Float32 | 30 | Input BP filter memory (23.85 kbit/s mode) |
|  | mem\_hf3 | Float32 | 30 | Input LP filter memory (23.85 kbit/s mode) |
|  | mem\_isp | Float32 | 16 | Old ISP vector |
|  | mem\_syn | Float32 | 16 | synthesis filter memory |
|  | mem\_syn2 | Float32 | 16 | modified synthesis memory |
|  | mem\_syn\_hf | Float32 | 16 | Higher band synthesis filter memory |
|  | mem\_isf | Float32 | 16 | Old ISF vector |
|  | mem\_hf\_wsp | Float32 | 9 | Open-loop lag gain filter memory |
|  | mem\_sig\_in | Float32 | 4 | Prefilter memory |
|  | mem\_sig\_out | Float32 | 4 | HP filter memory in the synthesis |
|  | mem\_hp400 | Float32 | 4 | HP filter memory |
|  | mem\_decim2 | Float32 | 3 | Open-loop LTP decimation filter memory |
|  | mem\_gp\_clip | Float32 | 2 | Memory of pitch clipping |
|  | mem\_preemph | Float32 | 1 | Preemphasis filter memory |
|  | mem\_deemph | Float32 | 1 | Deemphasis filter memory |
|  | mem\_wsp\_df | Float32 | 1 | Open-loop LTP deemphasis filter memory |
|  | mem\_w0 | Float32 | 1 | Weighting filter memory (applied to error signal) |
|  | mem\_ol\_gain | Float32 | 1 | Open-loop gain |
|  | mem\_ada\_w | Float32 | 1 | Weighting level depeding on open loop pitch gain |
|  | mem\_gc\_threshold | Float32 | 1 | Noise enhancer threshold |
|  | mem\_gain\_alpha | Float32 | 1 | Higher band gain weighting factor (23.85 kbit/s mode) |
|  | mem\_ol\_lag | Word32 | 5 | Open loop lag history |
|  | mem\_T0\_med | Word32 | 1 | Weighted open loop pitch lag |
|  | mem\_exc | Word16 | 505 | Excitation vector |
|  | mem\_isp\_q | Word16 | 16 | Old ISP vector |
|  | mem\_isf\_q | Word16 | 16 | Past quantized ISF prediction error |
|  | mem\_gain\_q | Word16 | 4 | Gain quantization memory |
|  | mem\_subfr\_q | Word16 | 4 | Scaling factor history |
|  | mem\_tilt\_code | Word16 | 1 | Preemhasis filter memory |
|  | mem\_q | Word16 | 1 | Old scaling factor |
|  | mem\_seed | Word16 | 1 | Random generation seed |
|  | \*vadSt | E\_DTX\_Vad\_State | 1 | See below in this table |
|  | \*dtx\_encSt | E\_DTX\_State | 1 | See below in this table |
|  | mem\_first\_frame | UWord8 | 1 | First frame indicator |
|  | mem\_ol\_wght\_flg | UWord8 | 1 | Switches lag weighting on and off |
|  | mem\_vad\_hist | UWord8 | 1 | VAD history |
| E\_DTX\_State | mem\_isf | Float32 | 128 | LSP history |
|  | mem\_distance | Float32 | 28 | ISF history distance matrix |
|  | mem\_distance\_sum | Float32 | 8 | Sum of ISF history distances |
|  | mem\_log\_en | Float32 | 8 | Logarithmic frame energy history |
|  | mem\_hist\_ptr | Word16 | 1 | Pointer to the cyclic history vectors |
|  | mem\_log\_en\_index | Word16 | 1 | Index for logarithmic energy |
|  | mem\_cng\_seed | Word16 | 1 | Comfort noise excitation seed |
|  | mem\_dtx\_hangover\_count | Word16 | 1 | DTX hangover period |
|  | mem\_dec\_ana\_elapsed\_count | Word16 | 1 | Counter for elapsed speech frames in DTX |
| E\_DTX\_Vad\_State | mem\_pow\_sum | Float64 | 1 | Power of previous frame |
|  | mem\_bckr\_est | Float32 | 12 | Background noise estimate |
|  | mem\_ave\_level | Float32 | 12 | Averaged input components for stationary estimation |
|  | mem\_leve | Float32 | 12 | Input levels of the previous frame |
|  | mem\_sub\_level | Float32 | 12 | Input levels calculated at the end of a frame (lookahead) |
|  | mem\_a\_data5 | Float32 | 10 | Memory for the filter bank |
|  | mem\_a\_data3 | Float32 | 6 | Memory for the filter bank |
|  | mem\_sp\_max | Float32 | 1 | Maximum level |
|  | mem\_speech\_level | Float32 | 1 | Estimated speech level |
|  | mem\_burst\_count | Word16 | 1 | Counts length of a speech burst |
|  | mem\_hang\_count | Word16 | 1 | Hangover counter |
|  | mem\_stat\_count | Word16 | 1 | Stationary counter |
|  | mem\_vadreg | Word16 | 1 | Flags for intermediate VAD decisions |
|  | mem\_pitch\_tone | Word16 | 1 | Flags for pitch and tone detection |
|  | mem\_sp\_est\_cnt | Word16 | 1 | Counter for speech level estimation |
|  | mem\_sp\_max\_cnt | Word16 | 1 | Counts frames that contains speech |

Table 6: Speech decoder static variables

| **Struct name** | **Variable** | **Type** | **Length** | **Description** |
| --- | --- | --- | --- | --- |
| Decoder\_State | mem\_gc\_thres | Word32 | 1 | Threshold for noise enhancer |
|  | mem\_exc | Word16 | 505 | INTERPOL]; /\* old excitation vector |
|  | mem\_isf\_buf | Word16 | 48 | ISF buffer(frequency domain) |
|  | mem\_hf | Word16 | 30 | HF band-pass filter memory |
|  | mem\_hf2 | Word16 | 30 | HF band-pass filter memory |
|  | mem\_hf3 | Word16 | 30 | HF band-pass filter memory |
|  | mem\_oversamp | Word16 | 24 | Synthesis oversampled filter memory |
|  | mem\_gain | Word16 | 23 | Gain decoder memory |
|  | mem\_syn\_hf | Word16 | 20 | HF synthesis memory |
|  | mem\_isp | Word16 | 16 | Old ISP (immittance spectral pairs) |
|  | mem\_isf | Word16 | 16 | Old ISF (frequency domain) |
|  | mem\_isf\_q | Word16 | 16 | Past ISF quantizer |
|  | mem\_syn\_hi | Word16 | 16 | Modified synthesis memory (MSB) |
|  | mem\_syn\_lo | Word16 | 16 | Modified synthesis memory (LSB) |
|  | mem\_ph\_disp | Word16 | 8 | Phase dispersion memory |
|  | mem\_sig\_out | Word16 | 6 | Hp50 filter memory for synthesis |
|  | mem\_hp400 | Word16 | 6 | Hp400 filter memory for synthesis |
|  | mem\_lag | Word16 | 5 | LTP lag history |
|  | mem\_subfr\_q | Word16 | 4 | Old maximum scaling factor |
|  | mem\_tilt\_code | Word16 | 1 | Tilt of code |
|  | mem\_q | Word16 | 1 | Old scaling factor |
|  | mem\_deemph | Word16 | 1 | Speech deemph filter memory |
|  | mem\_seed | Word16 | 1 | Random memory for frame erasure |
|  | mem\_seed2 | Word16 | 1 | Random memory for HF generation |
|  | mem\_seed3 | Word16 | 1 | Random memory for lag concealment |
|  | mem\_T0 | Word16 | 1 | Old pitch lag |
|  | mem\_T0\_frac | Word16 | 1 | Old pitch fraction lag |
|  | mem\_vad\_hist | UWord16 | 1 | VAD history |
|  | dtx\_decSt | D\_DTX\_State | 1 | See below in this table |
|  | mem\_bfi | UWord8 | 1 | Previous BFI |
|  | mem\_state | UWord8 | 1 | BGH state machine memory |
|  | mem\_first\_frame | UWord8 | 1 | First frame indicator |
| dtx\_decState | mem\_isf\_buf | Word16 | 128 | ISF vector history (8 frames) |
|  | mem\_isf | Word16 | 16 | ISF vector |
|  | mem\_isf\_prev | Word16 | 16 | Previous ISF vector |
|  | mem\_log\_en\_buf | Word16 | 8 | Logarithmic frame energy history |
|  | mem\_true\_sid\_period\_inv | Word16 | 1 | Inverse of true SID update rate |
|  | mem\_log\_en | Word16 | 1 | Logarithmic frame energy |
|  | mem\_log\_en\_prev | Word16 | 1 | Previous logarithmic frame energy |
|  | mem\_cng\_seed | Word16 | 1 | Comfort noise excitation seed |
|  | mem\_hist\_ptr | Word16 | 1 | Index to beginning of LSF history |
|  | mem\_dither\_seed | Word16 | 1 | Comfort noise dithering seed |
|  | mem\_cn\_dith | Word16 | 1 | Background noise stationarity information |
|  | mem\_since\_last\_sid | Word16 | 1 | Number of frames since last SID frame |
|  | mem\_dec\_ana\_elapsed\_count | UWord8 | 1 | Counts elapsed speech frames after DTX |
|  | mem\_dtx\_global\_state | UWord8 | 1 | DTX state flags |
|  | mem\_data\_updated | UWord8 | 1 | Flags CNI updates |
|  | mem\_dtx\_hangover\_count | UWord8 | 1 | Counts down in hangover period |
|  | mem\_sid\_frame | UWord8 | 1 | Flags SID frames |
|  | mem\_valid\_data | UWord8 | 1 | Flags SID frames containing valid data |
|  | mem\_dtx\_hangover\_added | UWord8 | 1 | Flags hangover period at end of speech |

# 5 Homing procedure

The principles of the homing procedures are described in [2]. The present document only includes a description of the 9 decoder homing frames. For each AMR-WB codec mode, the corresponding decoder homing frame has a fixed set of speech parameters. Table 7 shows the homing frame speech parameters for different modes.

Table 7: Table values for the decoder homing frame parameters for different modes

|  |  |
| --- | --- |
| **Mode** | **Speech Parameters** |
| 0 | 0, 49, 131, 84, 5, 50, 29, 2015, 8,0, 2061, 8,1, 3560, 8,0, 2981, 8 |
| 1 | 0, 49, 131, 55, 49, 38, 26, 29, 29, 3, 15, 7,15, 8, 16, 13, 7, 17,16, 8, 0, 16, 20, 16, 27, 8, 23,0, 27, 0, 27, 8 |
| 2 | 0, 49, 131, 55, 49, 38, 26, 29, 58, 1, 7, 63,127, 15, 70, 37, 1, 209, 210, 224, 96, 31, 7, 1, 256, 260, 271, 443, 31, 47, 0, 400, 238, 436, 347, 31 |
| 3 | 0, 49, 131, 55, 49, 38, 26, 29, 58, 1, 3847, 3845, 63, 127, 70, 34, 0, 3128, 4517, 192, 96, 0, 2, 1, 4160, 8036, 267, 443, 31, 46, 0, 3840, 7091, 432, 395, 31 |
| 4 | 0, 49, 131, 55, 49, 38, 26, 29, 58, 1, 3847, 3845, 3847, 3843, 70, 31, 0, 3648, 4764, 824, 2864, 0, 6, 1, 4160, 5220, 4319, 7131, 31, 47, 0, 112, 3764, 219, 211, 31 |
| 5 | 0, 49, 131, 55, 49, 38, 26, 29, 58, 1, 3, 2, 3, 2, 7223, 703, 7223, 703, 70, 0, 1, 3, 2, 2, 3, 9475, 9483, 3090, 8737, 0, 0, 1, 0, 0, 2, 0, 4112, 4400, 8415, 14047, 31, 38, 0, 2, 1, 3, 1, 91, 426, 13545, 12955, 0 |
| 6 | 0, 49, 131, 55, 49, 38, 26, 29, 58, 1, 161, 759, 3, 2, 127, 516, 6167, 447, 70, 11, 1, 264, 641, 2, 3, 123, 562, 8347, 4354, 0, 1, 1, 264, 408, 3, 0, 256, 308, 9487, 14047, 31, 46, 0, 320, 885, 2, 2, 464, 439, 11347, 12739, 0 |
| 7 | 0, 49, 131, 55, 49, 38, 26, 29, 58, 1, 1154, 1729, 1154, 1761, 447, 1519, 959, 495, 70, 27, 1, 1800, 1253, 665, 1960, 546, 164, 1043, 335, 0, 28, 1, 580, 196, 1187, 383, 1031, 1052, 359, 1531, 31, 45, 1, 1024, 893, 1272, 1920, 101, 876, 203, 1119, 31 |
| 8 | 0, 49, 131, 55, 49, 38, 26, 29, 58, 1, 1729, 1154, 1761, 1154, 1519, 959, 495, 447, 70, 3, 42, 1, 580, 1436, 1362, 1250, 901, 714, 24, 45, 0, 0, 0, 1, 68, 708, 1212, 383, 1048, 1611, 1756, 1467, 31, 1, 23, 0, 1536, 1460, 861, 1554, 410, 1368, 1008, 594, 31, 0 |

# 6 File formats

This clause describes the file formats used by the encoder and decoder programs. The test sequences defined in [1 also use the file formats described here.

## 6.1 Speech file (encoder input/decoder output)

Speech files read by the encoder and written by the decoder consist of 16-bit words where each word contains a 14-bit, left aligned speech sample. The byte order depends on the host architecture (e.g. MSByte first on SUN workstations, LSByte first on PCs etc.). Both the encoder and the decoder program process complete frames (of 320 samples) only.

This means that the encoder will only process *n* frames if the length of the input file is *n\*320 + k* words, while the files produced by the decoder will always have a length of *n\*320* words.

## 6.2 Mode control file (encoder input)

The encoder program can optionally read in a mode control file which specifies the encoding mode for each frame of speech processed. The file is a text file containing one number per speech frame. Each line contains one of the mode numbers 0-8.

## 6.3 Parameter bitstream file (encoder output/decoder input)

The files produced by the speech encoder/expected by the speech decoder are described in RFC 3267 [7], sections 5.1 and 5.3.

By using a preprocessor definition encoder/decoder can optionally use format described in TS26.201 that defines an octet-aligned frame format (Interface format 2) for the AMR-WB codec.

Annex A (informative):  
Change history

|  |  |  |  |  |  |  |  |
| --- | --- | --- | --- | --- | --- | --- | --- |
| **Change history** | | | | | | | |
| **Date** | **TSG #** | **TSG Doc.** | **CR** | **Rev** | **Subject/Comment** | **Old** | **New** |
| 2002-03 | 15 | SP-020073 |  |  | Presented at TSG SA#15 for approval | 2.0.0 | 5.0.0 |
| 2003-03 | 19 | SP-030090 | 001 | 1 | Correction to log(0) error in VAD decision with low SNR input signals | 5.0.0 | 5.1.0 |
| 2003-03 | 19 | SP-030090 | 002 | 1 | Correction to decoder with input of long sequence of NO\_DATA frames | 5.0.0 | 5.1.0 |
| 2003-03 | 19 | SP-030090 | 003 | 1 | Correction to "D\_UTIL\_pow2" function to be bitexact with TS26.173 counterpart | 5.0.0 | 5.1.0 |
| 2003-03 | 19 | SP-030090 | 004 | 1 | MMS compatible i/o format option | 5.0.0 | 5.1.0 |
| 2003-03 | 19 | SP-030090 | 005 |  | Correction for handling of RX\_NO\_DATA frames | 5.0.0 | 5.1.0 |
| 2003-03 | 19 | SP-030090 | 006 | 1 | Ambiguous expressions in the AMR-WB Floating-point C-Code | 5.0.0 | 5.1.0 |
| 2003-09 | 21 | SP-030447 | 008 |  | Possible decoder LPC coefficients overflow | 5.1.0 | 5.2.0 |
| 2004-12 | 26 | SP-040844 | 009 | 1 | Incorrect definition of vector nb\_of\_bits | 5.2.0 | 6.0.0 |
| 2007-03 | 35 | SP-070029 | 0011 |  | Maintaining bit-exactness with TS 26.173 after Correction in AMR decoder to avoid division by zero in RX-DTX Handling | 6.0.0 | 7.0.0 |
| 2007-03 | 35 | SP-070029 | 0012 |  | Bug fix to SID frame signaling in decoder | 6.0.0 | 7.0.0 |
| 2007-09 | 37 | SP-070626 | 0015 | 1 | Robust operation of AMRWB-decoder | 7.0.0 | 7.1.0 |
| 2008-12 | 42 |  |  |  | Version for Release 8 | 7.1.0 | 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 |
| 2012-09 | 57 |  |  |  | Version for Release 11 | 10.0.0 | 11.0.0 |
| 2014-09 | 65 |  |  |  | Version for Release 12 | 11.0.0 | 12.0.0 |
| 2015-03 | 67 | SP-150094 | 0016 | 1 | Correction on AMR-WB (noise energy initialization) | 12.0.0 | 12.1.0 |
| 2015-12 | 70 |  |  |  | Version for Release 13 | 12.1.0 | 13.0.0 |
| 2016-03 | 71 | SP-160077 | 0017 | 1 | Correction of AMR-WB | 13.0.0 | 13.1.0 |

|  |  |  |  |  |  |  |  |
| --- | --- | --- | --- | --- | --- | --- | --- |
| **Change history** | | | | | | | |
| **Date** | **Meeting** | **TDoc** | **CR** | **Rev** | **Cat** | **Subject/Comment** | **New version** |
| 2017-03 | 75 |  |  |  |  | Version for Release 14 | 14.0.0 |
| 2017-06 | 76 | SP-170329 | 0018 | - | F | Correction to prevent undefined behaviour | 14.1.0 |
| 2018-06 | 80 |  |  | - |  | Version for Release 15 | 15.0.0 |
| 2018-12 | 82 | SP-180981 | 0020 | - | F | Corrections to AMR-WB floating-point for 64-bit systems | 16.0.0 |