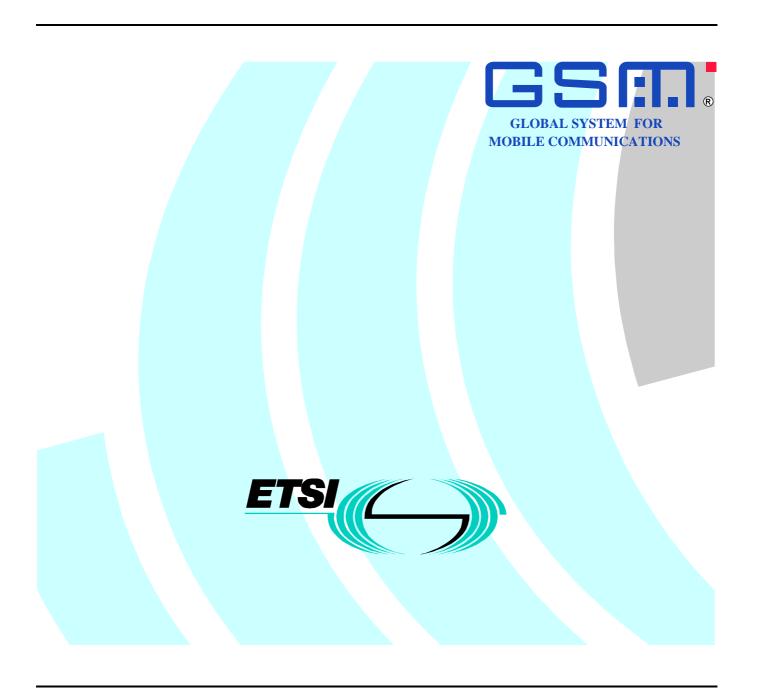
Draft ETSI EN 300 727 V7.0.0 (1999-08)

European Standard (Telecommunications series)

Digital cellular telecommunications system (Phase 2+); Substitution and muting of lost frames for Enhanced Full Rate (EFR) speech traffic channels (GSM 06.61 version 7.0.0 Release 1998)



Reference

REN/SMG-020661Q7 (6qc03i00.PDF)

Keywords

Digital cellular telecommunications system, Global System for Mobile communications (GSM)

ETSI

Postal address

F-06921 Sophia Antipolis Cedex - FRANCE

Office address

650 Route des Lucioles - Sophia Antipolis Valbonne - 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

Internet

secretariat@etsi.fr
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Foreword

This European Standard (Telecommunications series) has been produced by ETSI Technical Committee Special Mobile Group (SMG), and is now submitted for the ETSI standards One-step Approval Procedure.

The present document defines a frame substitution and muting procedure which is used by the Receive (RX) Discontinuous Transmission (DTX) handler when one or more lost speech or Silence Descriptor (SID) frames are received from the Radio Sub System (RSS) within the digital cellular telecommunications system.

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

Version 7.x.y

where:

- 7 indicates Release 1998 of GSM Phase 2+
- x the second digit is incremented for all changes of substance, i.e. technical enhancements, corrections, updates, etc
- y the third digit is incremented when editorial only changes have been incorporated in the specification.

Proposed national transposition dates					
Date of latest announcement of this EN (doa):	3 months after ETSI publication				
Date of latest publication of new National Standard or endorsement of this EN (dop/e):	6 months after doa				
Date of withdrawal of any conflicting National Standard (dow):	6 months after doa				

1 Scope

The present document defines a frame substitution and muting procedure which shall be used by the Receive (RX) Discontinuous Transmission (DTX) handler when one or more lost speech or Silence Descriptor (SID) frames are received from the Radio Sub System (RSS).

The requirements of the present document are mandatory for implementation in all GSM Base Station Systems (BSS)s and Mobile Stations (MS)s capable of supporting the enhanced Full Rate speech traffic channel. It is not mandatory to follow the bit exact implementation outlined in the present document and the corresponding C-source code.

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.
- A non-specific reference to an ETS shall also be taken to refer to later versions published as an EN with the same number.
- For this Release 1998 document, references to GSM documents are for Release 1998 versions (version 7.x.y).
- [1] GSM 05.03: "Digital cellular telecommunications system (Phase 2+); Channel coding".
- [2] GSM 06.60: "Digital cellular telecommunications system (Phase 2+); Enhanced Full Rate (EFR) speech transcoding".
- [3] GSM 06.81: "Digital cellular telecommunications system (Phase 2+); Discontinuous transmission (DTX) for Enhanced Full Rate (EFR) speech traffic channels".
- [4] GSM 08.60: "Digital cellular telecommunications system (Phase 2+); Inband control of remote transcoders and rate adaptors for Enhanced Full Rate (EFR) and full rate traffic channels".

3 Definitions and abbreviations

3.1 Definitions

For the purposes of the present document, the following definition applies:

5-point median operation: Consists of sorting the 5 elements belonging to the set for which the median operation is to be performed in an ascending order according to their values, and selecting the third largest value of the sorted set as the median value.

Further definitions of terms used in the present document can be found in GSM 06.60 [2], GSM 06.81 [3], GSM 05.03 [1] and GSM 08.60 [4].

3.2 Abbreviations

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

BFI Bad Frame Indication from Radio Sub System

BSI_Abis Bad Sub-block Indication obtained from A-bis CRC checks

PrevBFI Bad Frame Indication of Previous frame

RSS Radio Sub System

RX Receive

DTX Discontinuous transmission
SID SIlence Descriptor frame
CRC Cyclic Redundancy Check
CCU Channel Coding Unit
median5 5-point median operation

TRAU Transcoding Rate Adaptation Unit

4 General

The purpose of frame substitution is to conceal the effect of lost frames. The purpose of muting the output in the case of several lost frames is to indicate the breakdown of the channel to the user and to avoid generating possible annoying sounds as a result from frame substitution procedure.

The RSS indicates lost speech or SID frames by setting its Bad Frame Indication flag (BFI) based on its 3-bit and 8-bit CRCs and possibly other error detection mechanisms. The TRAU calculates from the CRCs inserted by the CCU in the TRAU frames one BSI_Abis flag for every sub-block of speech parameters. If either one or more of these flags is set, the speech decoder shall either perform frame substitution or subframe substitution.

The example solution provided in clause 6 applies only for bad frame handling on a complete speech frame basis. However some parts could be modified for substitution of bad sub-blocks.

5 Requirements

5.1 Error detection

An error is detected and the BFI-flag is set-by the RSS according to the principle described in clause 4.

5.2 Lost speech frames

Normal decoding of lost speech frames would result in very unpleasant noise effects. In order to improve the subjective quality, lost speech frames shall be substituted with either a repetition or an extrapolation of the previous good speech frame(s). This substitution is done so that it will gradually decrease the output level, resulting in silencing of the output. Subclause 6.1 gives an example solution.

5.3 First lost SID frame

A single lost SID frame shall be substituted by the last valid SID frame and the procedure for valid SID frames be applied as described in GSM 06.81 [3].

5.4 Subsequent lost SID frames

For the second lost SID frame, a muting technique shall be used on the comfort noise that will gradually decrease the output level (-3 dB/frame), resulting in silencing of the output of the decoder.

For subsequent lost SID frames, the muting of the output shall be maintained. Subclause 6.2 gives an example solution.

6 Example solution

The C-code of the following example is embedded in the bit exact software of the enhanced full rate codec.

6.1 Example solution for substitution and muting of lost speech frames

This example solution for substitution and muting is based on a state machine with seven states (figure 1).

The system starts in state 0. Each time a bad frame is detected, the state counter is incremented by one and is saturated when it reaches 6. Each time a good speech frame is detected, the state counter is reset to zero, except when we are in state 6, where we set the state counter to 5. The state indicates the quality of the channel: the bigger the state counter, the worse the channel quality is. The control flow of the state machine can be described with the following C-code (**BFI** = bad frame indicator, **State** = state variable):

In addition to this state machine, the **Bad Frame Flag** from the previous frame is checked (**PrevBFI**). The processing depends on the value of the **State**-variable. In states 0 and 5, the processing depends also on the two flags **BFI** and **PrevBFI**.

The procedure can be described as follows:

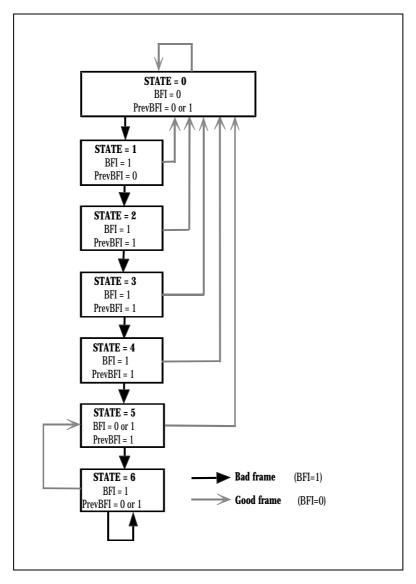


Figure 1: State machine for controlling the bad frame substitution

BFI = 0, PrevBFI = 0, State = 0

No error is detected in the received or in the previous received speech frame. The received speech parameters are used normally in the speech synthesis. The current frame of speech parameters is saved.

BFI = 0, PrevBFI = 1, State = 0 or 5

No error is detected in the received speech frame but the previous received speech frame was bad. The LTP-gain and fixed codebook gain are limited below the values used for the last received good subframe:

$$g^{p} = \begin{cases} g^{p}, & g^{p} \leq g^{p}(-1) \\ g^{p}(-1), & g^{p} > g^{p}(-1) \end{cases}$$
 (1)

where g^p = current decoded LTP-gain, $g^p(-1)$ = LTP-gain used for the last good subframe (BFI = 0), and

$$g^{c} = \begin{cases} g^{c}, & g^{c} \leq g^{c}(-1) \\ g^{c}(-1), & g^{c} > g^{c}(-1) \end{cases}$$
 (2)

where g^c = current decoded fixed codebook-gain and $g^c(-1)$ = fixed codebook gain used for the last good subframe (BFI = 0).

The rest of the received speech parameters are used normally in the speech synthesis. The current frame of speech parameters is saved.

BFI = 1, PrevBFI = 0 or 1, State = 1...6

An error is detected in the received speech frame and the substitution and muting procedure is started. The LTP-gain and fixed codebook gain are replaced by attenuated values from the previous subframes:

$$g^{p} = \begin{cases} P(state) \ g^{p}(-1), & g^{p}(-1) \leq median5(g^{p}(-1), ..., g^{p}(-5)) \\ P(state) \ median5(g^{p}(-1), ..., g^{p}(-5)), & g^{p}(-1) > median5(g^{p}(-1), ..., g^{p}(-5)) \end{cases}$$
(3)

where g^p = current decoded LTP-gain, g^p (-1),..., g^p (-n) = LTP-gains used for the last n subframes, median5() = 5-point median operation, P(state) =attenuation factor (P(1) = 0.98, P(2) = 0.98, P(3) = 0.8, P(4) = 0.3, P(5) = 0.2, P(6) = 0.2), state =state number, and

$$g^{c} = \begin{cases} C(state) \ g^{c}(-1), & g^{c}(-1) \leq median5(g^{c}(-1), ..., g^{c}(-5)) \\ C(state) \ median5(g^{c}(-1), ..., g^{c}(-5)), & g^{c}(-1) > median5(g^{c}(-1), ..., g^{c}(-5)) \end{cases}$$
(4)

where g^c = current decoded fixed codebook gain, $g^c(-1), \ldots, g^c(-n)$ = fixed codebook gains used for the last n subframes, median5() = 5-point median operation, C(state) = attenuation factor (C(1) = 0.98, C(2) = 0.98, C(3) = 0.98, C(4) = 0.98, C(5) = 0.98, C(6) = 0.7), and state = state number.

The higher the state value is, the more the gains are attenuated. Also the memory of the predictive fixed codebook gain is updated by using the average value of the past four values in the memory:

$$ener(0) = \frac{1}{4} \sum_{i=1}^{4} ener(-i)$$
 (5)

The past LSFs are used by shifting their values towards their mean:

$$lsf_q(i) = lsf_q(i) = \alpha \ past_lsf_q(i) + (1 - \alpha)mean_lsf(i), \quad i = 0...9$$
 (6)

where $\alpha = 0.95$, lsf_q1 and lsf_q2 are two sets of LSF-vectors for current frame, $past_lsf_q$ is lsf_q2 from the previous frame, and $mean_lsf$ is the average LSF-vector.

The LTP-lag values are replaced by the past value from the 4th subframe of the previous frame.

The received fixed codebook excitation pulses from the erroneous frame are always used as such.

6.2 Example solution for substitution and muting of lost SID frames

The first lost SID frame is replaced by the last valid SID frame.

For subsequent lost SID frames, the last valid SID frame is repeated, but the fixed codebook gain is decreased with a constant value of -3 dB in each frame down to the minimum value of 0. This value is maintained if additional lost SID frames occur.

Annex A (informative): Change Request History

Change history							
SMG No.	TDoc. No.	CR. No.	Section affected	New version	Subject/Comments		
SMG#22				4.0.1	ETSI Publication		
SMG#20				5.1.2	Release 1996 version		
SMG#27				6.0.0	Release 1997 version		
SMG#29				7.0.0	Release 1998 version		

History

Document history									
V7.0.0	August 1999	One-step Approval Procedure	OAP 9956:	1999-08-25 to 1999-12-24					