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(GSM 06.11 version 5.0.1)

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Contents

Forev	ord	Ę
1	Scope	7
2	Normative references	
3	Definitions and abbreviations	7
4	General	7
5	Requirements	7
	5.1 First lost speech frame	7
	5.2 Subsequent lost speech frames	7
	5.3 First lost SID frame	7
	5.4 Subsequent lost SID frame	8
6	Example solution	8
Histor	<i>y</i> 1	(

Page 4 ETS 300 962 (GSM 06.11 version 5.0.1): May 1997

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Foreword

This European Telecommunication Standard (ETS) has been produced by the Special Mobile Group (SMG) Technical Committee (TC) of the European Telecommunications Standards Institute (ETSI).

This ETS specifies the substitution and muting of lost frames for full rate speech channels for the digital cellular telecommunications system.

The specification from which this ETS has been derived was originally based on CEPT documentation, hence the presentation of this ETS may not be entirely in accordance with the ETSI/PNE Rules.

Transposition dates				
Date of adoption:	18 April 1997			
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Page 6 ETS 300 962 (GSM 06.11 version 5.0.1): May 1997

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1 Scope

[1]

This European Telecommunication Standard (ETS) defines a frame substitution and muting procedure which shall be used by the RX DTX handler when one or more lost speech or SID frames are received from the radio subsystem.

The requirements of this ETS are mandatory for implementation in all GSM Base Station Systems (BSS) and Mobile Stations (MS).

2 Normative references

This ETS incorporates by dated and undated reference, provisions from other publications. These normative references are cited at the appropriate places in the text and the publications are listed hereafter. For dated references, subsequent amendments to or revisions of any of these publications apply to this ETS only when incorporated in it by amendment or revision. For undated references, the latest edition of the publication referred to applies.

	Abbreviations and acronyms".
[2]	GSM 06.10 (ETS 300 961): "Digital cellular telecommunications system; Full

GSM 01.04 (ETR 350): "Digital cellular telecommunications system (Phase 2+);

rate speech; Transcoding".

[3] GSM 06.31 (ETS 300 964): "Digital cellular telecommunications system; Full rate speech; Discontinuous Transmission (DTX) for full rate speech traffic channels".

3 Definitions and abbreviations

The definitions of terms used in this ETS can be found in GSM 06.31 [3].

Abbreviations used in this specification are listed in GSM 01.04 [1].

4 General

The purpose of the 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.

5 Requirements

5.1 First lost speech frame

Normal decoding of lost speech frames would result in very unpleasant noise effects. In order to improve the subjective quality, the first lost speech frame shall be substituted with either a repetition or an extrapolation of the previous good speech frame(s). Lost speech frames shall not be delivered to the speech decoder, nor shall the output be muted directly.

5.2 Subsequent lost speech frames

For subsequent lost speech frames, a muting technique shall be used that will gradually decrease the output level, resulting in silencing of the output after a maximum of 320 ms. Clause 6 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 06.31 [3].

5.4 Subsequent lost SID frame

For the second lost SID frame, a muting technique shall be used on the comfort noise that will gradually decrease the output level, resulting in silencing of the output after a maximum of 320 ms. Clause 6 gives an example solution.

For subsequent lost SID frames, the muting of the output shall be maintained.

6 Example solution

For guidance, an example solution is given.

The first lost speech frame is replaced at the speech decoder input by the previous good speech frame. Normal decoding is then performed.

The muting procedure to be used in the case of subsequent lost speech frames or for comfort noise frames following the second lost SID frame is as follows:

The pseudo-logarithmic encoded block amplitude Xmaxcr (GSM 06.10 [2]), coded on the interval from 0 to 63, is decreased with a constant value d=4 in each frame, down to the lowest possible value. Consequently, Xmaxcr will be reduced gradually, and the output muted after a maximum of 320 ms. The grid position parameters are chosen randomly between 0 and 3 during this time.

For subsequent unusable frames, after the frame where Xmaxcr reached the lowest possible value, "silence frames" are passed from the RX DTX handler to the speech decoder to guarantee a low output level under all conditions. The silence frame is defined in table 1.

Table 1: Encoded parameters (GSM 06.10) of the silence frame

Log area ratio 1 = 42
Log area ratio 2 = 39
Log area ratio 3 = 21
Log area ratio 4 = 10
Log area ratio 5 = 9
Log area ratio 6 = 4
Log area ratio 7 = 3
Log area ratio 8 = 2

LTP gain = 0LTP lag = 40

Grid position = 1Block amplitude = 0

RPE pulse no. 1 = 3RPE pulse no. 2 = 4RPE pulse no. 3 = 3RPE pulse no. 4 = 4RPE pulse no. 5 = 4

RPE pulse no. 6 = 3RPE pulse no. 7 = 3RPE pulse no. 8 = 3RPE pulse no. 9 = 3RPE pulse no. 10 = 4RPE pulse no. 11 = 4

RPE pulse no. 12 = 3RPE pulse no. 13 = 3 - repeated for each subsegment

History

Document history							
December 1996	Unified Approval Procedure	UAP 61:	1996-12-16 to 1997-04-11				
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