

STATISTICAL TIME DIVISION MULTIPLEXING ARCHITECTURES AND DESIGN

A2

15 mV

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200mV

20mV



0.1 500ns

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21. Concept of Frames Discarding and Multiplexing efficiency

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21.0 Abstract

At low bit rates speech is coded frame by frame, each frame of a size of 20-30ms. Perceptually, each frame may be different, depends upon the speech signal properties (voiced or unvoiced). The frames with lower perception may be one of the potential candidates to be dropped and reconstructed to maximize the bandwidth of the systems. This chapter explains to how these frames might be identified and used for STDM designs.

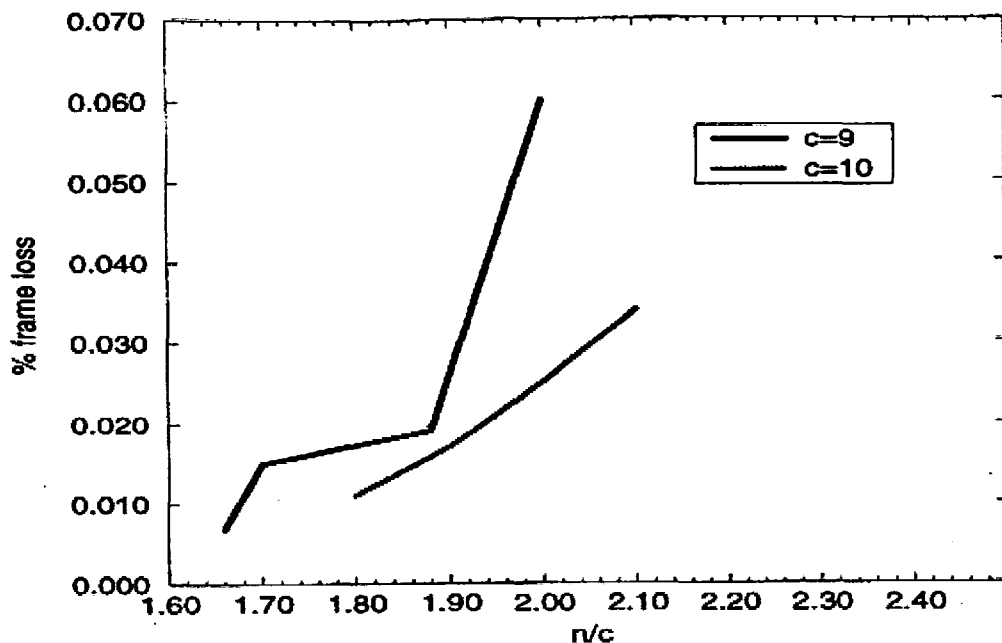


Figure 21.0-1: DSI advantage for conversational speech

In statistical multiplexing users are basically allocated bandwidth less than their actual peak requirements, that is, they are allowed bandwidth for transmission during talkspurts. As the number of users are increased higher than the output link capacity. The simultaneous talkspurts from e users are also increased. During higher activities, some user's talkspurts have to be