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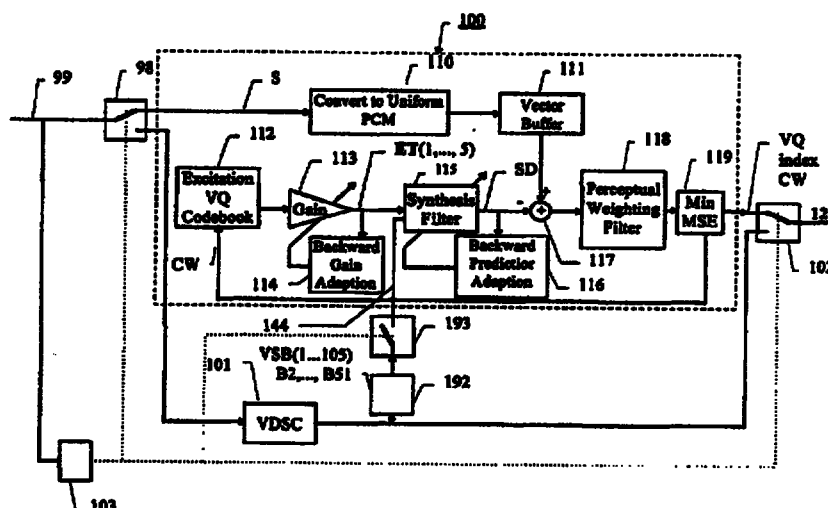
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(54) Title: METHOD AND APPARATUS IN CODING DIGITAL INFORMATION



(57) Abstract

A speech encoder (100) receives speech signals (S) which are encoded and transmitted on a communication channel (120). Silence in the speech is utilized by a data encoder (101) to transmit data on the speech frequency band via the channel (120). A signal classifier adaptor (116), gain predictor (113, 114) and excitation codebook (112). The speech encoder (100) has synthesis filter (115) with state variables in a delay line, predictor and updated in a buffer (192). On switching (103, 102, 193) from data to speech, the buffer state variables are fed into the synthesis filter delay line via an input (144) for smooth transition in the speech encoding. Coefficient values in the synthesis filter (115) and an excitation signal (ET(1...5)) are generated. Thereby a buffer in the gain predictor (113, 114) is preset and its predictor coefficients and gain are generated. The incoming speech signal (S) newly detected is encoded (CW) by the values generated in the speech encoder (100), which is successively adapted. The receiver side has corresponding speech and data decoders.