



## A NEW CODING METHOD FOR SPEECH AND AUDIO SIGNALS (WedPmOR1)

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### ★ Abstract :

In this paper a new representation or modeling method of speech signals is introduced. The proposed method is based on the generation of the so-called Predefined Signature  $S=\{SR\}$  and Envelope vector  $E=\{EK\}$  Sets (PSEVS). These vector sets are speaker and language independent. In this method, once the speech signals are divided into frames with selected lengths, then each frame signal piece  $X_i$  is reconstructed by means of the mathematical form of  $X_i=C_iEKSR$ . In this representation,  $C_i$  is called the frame coefficient,  $SR$  and  $EK$  are the vectors properly assigned from the PSEVS respectively. It is shown that the proposed method provides fast reconstruction and substantial compression ratio with acceptable hearing quality.