TELEPHONE SPEECH RECOGNITI ON US I NG NEURAL NETWORKS AND HI DDEN MARKOV MODELS

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ABSTRACT

The performance of well-trained speech recognizers using high quality full bandwidth speech data is usually deusedin real world environment

the testing and the training environments. Since the low quality speech feature vectors are transformed to high quality ones, it

can outperform the retrained recognizer that is train quality speech. Therefore, the papproach is

monophone HMM with 30 Gaussian distributions per state. In
total, the system has 3,630 Gaussian distributions. 1,344 utterances from NTI MI T² test data are used for testing. When th
system is trained and tested under the sa
both using TI MI T co

a system which is comparable to the retrained recognizer, but
with much less training data.
The advantages of this approach are as follows. First, it does
not require retraining of the speech recognizer
task in terms of trainin