



Tandem acoustic modeling in large-vocabulary recognition

Dan Ellis • Columbia University & ICSI • dpwe@ee.columbia.edu
 Rita Singh • Carnegie Mellon University • rsingh@cs.cmu.edu
 Sunil Sivadas • Oregon Graduate Institute • sunil@ece.ogi.edu

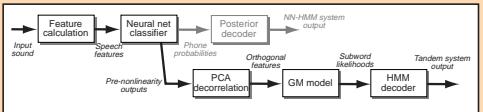
Carnegie
Mellon



Summary: In tandem acoustic modeling, classification is performed by a neural net followed by a Gaussian mixture model, achieving dramatic improvements on small-vocabulary tasks. For the larger SPINE1 task, much of the benefit disappears when used with context-dependent modeling and MLLR adaptation.

Introduction

- Tandem acoustic modeling refers to using the outputs of a discriminantly-trained neural network as the inputs to a conventional GMM-HMM speech recognizer. Two acoustic models, neural net and Gaussian mixture, are thus used in tandem:



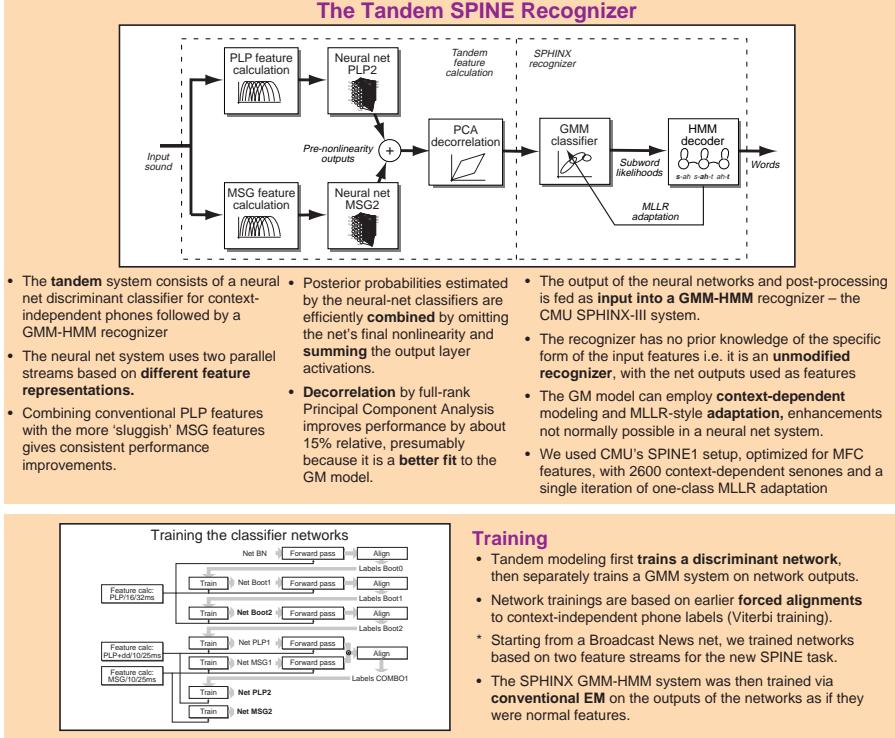
- When working with the ETSI Aurora noisy digits task, the tandem architecture, in conjunction with posterior-level feature stream combination facilitated WER reductions of over 50%:

Aurora results	WER% / SNR		WER ratio%
Feature	Clean	15 dB	5 dB
GMM MFC baseline	1.4	3.7	15.9
NN MFC baseline	1.6	2.6	8.7
Tandem MFC	0.9	2.1	8.0
Tandem PLP+MSG	0.7	1.5	7.2
47.2			

- We wanted to see if these kinds of improvements could be extended to tasks involving larger vocabularies and more speech variation. We therefore applied the same techniques to the SPINE1 task.

The SPINE1 task

- The first Speech In Noisy Environments task (SPINE1) was defined by the Naval Research Laboratory (NRL). An evaluation was conducted in August 2000.
- The SPINE1 task consists of dialogs between speakers in separate booths engaged in a game of 'Battleships'. Various pre-recorded noises are played in the booths to simulate real-world conditions.
- The task has a vocabulary of about 5,000 words, with natural and informal grammar and pronunciation.
- About 8 hours of transcribed training material, in a range of background noise conditions, was made available.
- This task is very challenging: In the evaluation, the best performance (from a combination of systems) was around 26% WER.



Results

- We compared 4 feature sets:
 - mfc - standard MFC features
 - plp - comparable PLP features
 - tandem1 - Tandem based on PLP
 - tandem2 - Tandem with PLP+MSG
 in 3 HMM model conditions:
 - CI - 39 context-indep. phone states
 - CD - 2600 context-dep. senone states
 - CD+MLLR - added MLLR adaptation
- For the **Context Independent** models, the tandem2 features reduced the baseline WER by 31%.
- Moving to **Context Dependent** models effects much larger improvements on the regular features (mfc, plp) than on the tandem features, bringing all results close together.
- Adding **MLLR adaptation** benefits the tandem systems slightly more, making the tandem2 system the best by a small margin.

Discussion

- Neural nets (discriminant) followed by GMMS (distribution models) work well for modeling **context-independent phones** even for natural, unconstrained speech.
- Tandem features **interact poorly** with **context-dependent** state models. Perhaps the context-independent network outputs are confounding the contextual cues within each class.
- MLLR benefits tandem** CD systems more than conventional features: contextual information may be more variable (but still present) in tandem features.

Future work

- Would a larger set of **context-dependent discriminant classes** (perhaps a factored network) work better?
- How does performance depend on **training set size**? Should the nets and GMMS be trained on separate data?
- What is the effect of additional processing (normalization, deltas) in the **posterior-features domain**?
- Would it help to **train the net** to a more directly relevant criterion?

