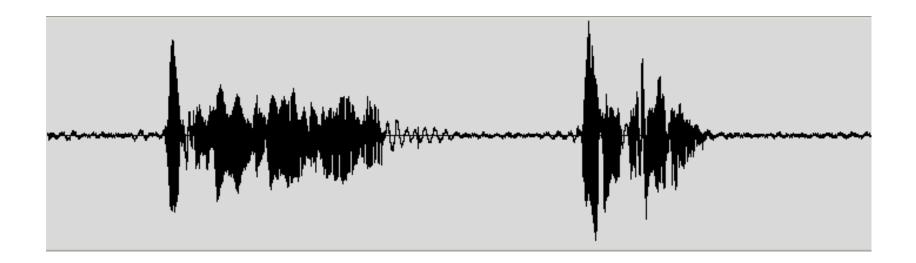
Context-Dependent Pre-Trained Deep Neural Networks for Large-Vocabulary Speech Recognition Paul Hensch







Large-Vocabulary Speech Recognition



Complications

Speaker

Accents
Dialect
Style
Emotion
Coarticulation
Reduction
Pronunciation
Hesitation

Environment

Noise Side talk Reverberation

Device

Head phone Land phone Speaker phone Cell phone



Large-Vocabulary Speech Recognition



Hidden Markov Models

Find Phones, which matches input

Neural networks

Look on frequency changes over time to recognize

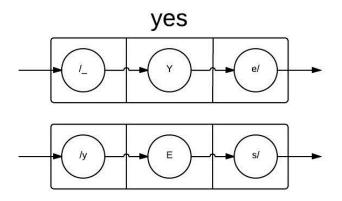
Hybrid

- CD-GMM-HMM
- CD-DNN-HMM





- Similar classes of sounds, or phones
 - Diphone
 - Triphone
 - Quinphone
- Senons are used as the DNN output unit





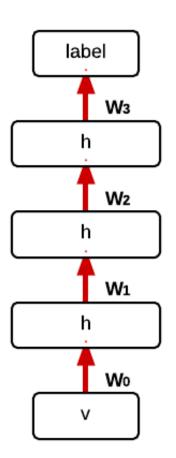


Deep Belief Network

(not dynamic Bayes net)

2 Stages:

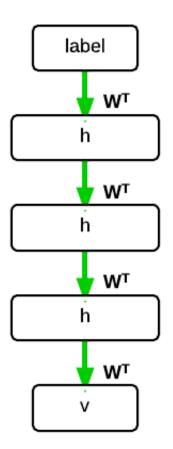
- Pre-training
- Fine-tuning





Pre-training advantages:

- Often also achieve lower training error
- Sort of data-dependent regularization

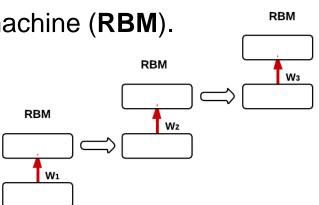


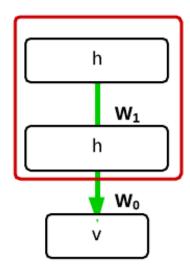


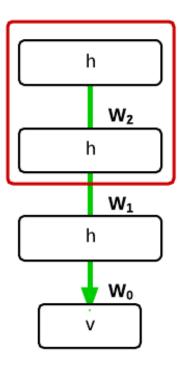


Pre-training

learning the connection weights in a DBN that is equivalent to training each adjacent pair of layers as an restricted Boltzmann machine (**RBM**).











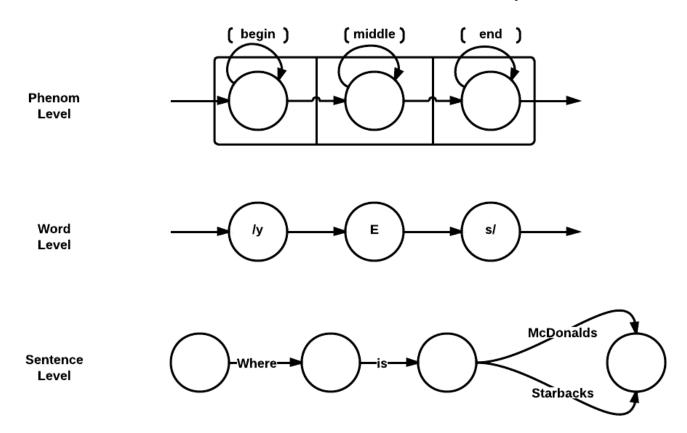
Last steps of training:

With pre-training complete, add a randomly initialized softmax output layer and use backpropagation to fine-tune all the weights.

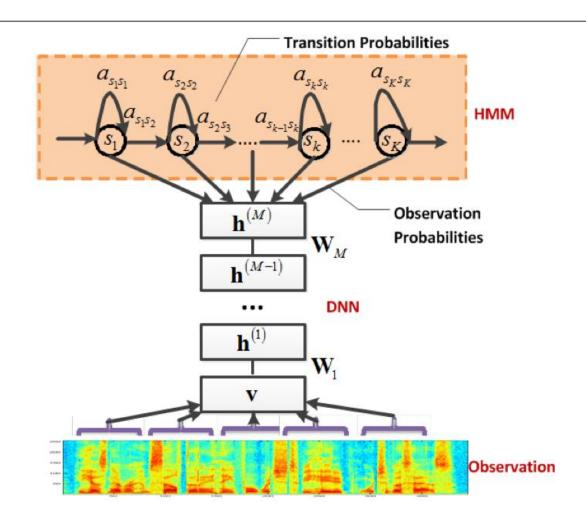




Hidden Markov Models is the dominant technique for LVSR











More advantages:

- Implement a CD-DNN-HMM system with only minimal modifications to an existing CD-GMM-HMM system
- Any improvements in modeling units that are incorporated into the CD-GMM-HMM baseline system, such as cross-word triphone models, will be accessible to the DNN





Business search dataset collected from the Bing mobile voice search application

- Collected under real usage scenarios in 2008
- Sampled at 8 kHz
- Encoded with the GSM codec
- contains all kinds of variations: noise, music, sidespeech, accents, sloppy pronunciation ...





INFORMATION ON THE BUSINESS SEARCH DATASET

	Hours	Number of Utterances
Training Set	24	32,057
Development Set	6.5	8,777
Test Set	9.5	12,758

- Dataset contains 65 K word unigrams, 3.2 million word bigrams, and 1.5 million word tri-grams.
- Sentence length is 2.1 tokens





Compare sentences (sentence accuracy) instead of word accuracy.

G. Zweig and P. Nguyen, "A segmental CRF approach to large vocabulary continuous speech recognition"

Difficulties:

- "Mc-Donalds" "McDonalds"
- "Walmart" "Wal-mart"
- "7-eleven" "7 eleven" "seven-eleven."

Maximum of 94% accuracy.





Baseline Systems

- trained clustered cross-word triphone GMM-HMM
- The performance of the best CD-GMM-HMM configuration is summarized in Table

maximum-likelihood
maximum mutual information
minimum phone error

	Criterion	Dev Accuracy	Test Accuracy
	ML	62.9%	60.4%
ار	MMI	65.1%	62.8%
	MPE	65.5%	63.8%





CD-DNN-HMM Results and Analysis

COMPARISON OF CONTEXT-INDEPENDENT MONOPHONE STATE LABELS AND CONTEXT-DEPENDENT TRIPHONE SENONE LABELS

# Hidden	# Hidden	Label	Dev
Layers	Units	Type	Accuracy
1	2K Monophone Sta		59.3%
1	2K	Triphone Senones	68.1%
3	2K	Monophone States	64.2%
3	2K	Triphone Senones	69.6%





CD-DNN-HMM Results and Analysis

CONTEXT-DEPENDENT MODELS WITH AND WITHOUT PRE-TRAINING

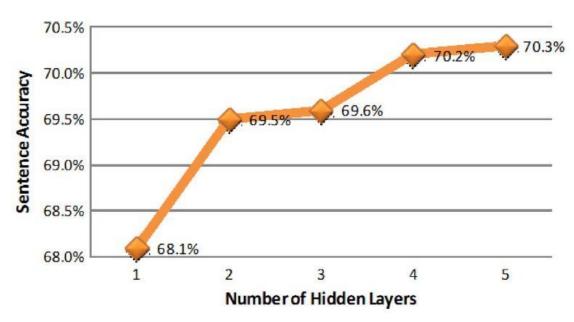
Model	# Hidden	# Hidden	Dev
Type	Layers	Units	Accuracy
without pre-training	1	2K	68.0%
without pre-training	2	2K	68.2%
with pre-training	1	2K	68.1%
with pre-training	2	2K	69.5%





CD-DNN-HMM Results and Analysis

Relationship between the recognition accuracy and the number of layers. Context-dependent models with 2 K hidden units







Training and Decoding Time

- Trainer written in Python
- Carried out on Dell Precision T3500 workstation
 - Quad core computer
 - CPU clock speed of 2.66 GHz, 8 MB of L3 CPU cache
 - 12 GB of 1066 MHz DDR3 SDRAM.
 - NVIDIA Tesla C1060 (GPGPU), which contains 4 GB of GDDR3 RAM and 240 processing cores





Training and Decoding Time

Summary of Training Time Using 24 Hours of Training Data and 2 K Hidden Units Per Layer

Type	# of Layers	Time Per Epoch	# of Epochs	
Pre-train	1	0.2 h	50	
Pre-train	2	0.5 h	20	
Pre-train 3 Pre-train 4		0.6 h	20	
		0.7 h	20	
Pre-train	5	0.8 h	20	
Fine-tune	4	1.2 h	12	
Fine-tune	5	1.4 h	12	





Training and Decoding Time

Processing	# of	DNN Time	Search Time	Real-time
Unit	Layers	Per Frame	Per Frame	Factor
CPU	4	4.3 ms	1.5 ms	0.58
GPU	4	0.16 ms	1.5 ms	0.17
CPU	5	5.2 ms	1.5 ms	0.67
GPU	5	0.20 ms	1.5 ms	0.17

• five-layer CD-DNN-HMM, pre-training takes

$$0.2 \times 50 + 0.5 \times 20 + 0.6 \times 20 + 0.7 \times 20 + 0.8 \times 20 = 62$$
 hours

Fine-tuning takes

$$1.4 \times 12 = 16.8 \text{ hours}$$



Training and Decoding Time

Observations:

The bottleneck in the training process is the mini-batch stochastic gradient descend (SGD) algorithm used to train the DNNs

It is extrapolated that using similar technics described in this paper, it should be possible to train an effective CD-DNN-HMM system that exploits **2000 hours** of training data in about **50 days**



CONCLUSION AND FUTURE WORK



- CD-DNN-HMM is more expensive than GMM
- CD-DNN-HMM performs better than GMM
- Finding new ways to parallelize training
- Finding highly effective speaker and environment adaptation algorithms for DNN-HMMs
- The training in this study used the embedded Viterbi algorithm, which is not optimal (MFCC)



Thank your for your attention



Questions?..

