Microsoft Research

Deep Learning: Speech & Information Processing

New-Generation Models & Methodology for Advancing Speech Technology and Information Processing

Li Deng

Microsoft Research, Redmond, USA

CCF, Beijing, July 8, 2013

(including joint work with colleagues at MSR, U of Toronto, etc.)

Microsoft Research

Outline

PART I: Basics of Deep Learning (DL)

--- including impact and recent history of DL (Deep Neural Net, DNN) in speech recognition

PART II: Deeper Substance of DL

--- including connections to other ML paradigms, examples of incorporating speech knowledge in DL architecture, and recent experiments in speech recognition

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Deep Learning (DL) Basics

1. **Deep Learning (aka Deep Structured Learning, Hierarchical Learning):** a class of machine learning techniques, where many layers of information processing stages in hierarchical architectures are exploited for unsupervised feature learning and for pattern analysis/classification.
2. **Deep belief nets (DBN):** probabilistic generative models composed of multiple layers of stochastic, hidden variables. The top two layers have undirected, symmetric connections between them. The lower layers receive top-down, directed connections from the layer above. (key: stacked RBMs; Hinton: **Science, 2006**)
3. **Boltzmann machine (BM)**: a network of symmetrically connected, neuron-like units that make stochastic decisions about whether to be on or off.
4. **Restricted Boltzmann machine (RBM):** a special BM consisting of a layer of visible units and a layer of hidden units with no visible-visible or hidden-hidden connections. (Key: contrastive divergence learning)
5. **Deep neural nets (DNN, or "DBN" before Nov 2012):** multilayer perceptrons with many hidden layers, whose weights are often initialized (pre-trained) using stacked RBMs or DBN (DBN-DNN) or discriminative pre-training.
6. **Deep auto-encoder:** a DNN whose output is the data input itself, often pre-trained with DBN (Deng/Hinton, interspeech 2010; Hinton, Science 2006)
7. **Distributed representation**: a representation of the observed data in such a way that they are modeled as being generated by the interactions of many hidden factors. A particular factor learned from configurations of other factors can often generalize well. Distributed representations form the basis of deep learning.

More on "Deep Learning"

- ***Definition 1****:* A class of machine learning techniques that exploit many layers of non-linear information processing for supervised or unsupervised feature extraction and transformation, and for pattern analysis and classification.

- ***Definition 2***: "A sub-field within machine learning that is based on algorithms for learning multiple levels of representation in order to model complex relationships among data. Higher-level features and concepts are thus defined in terms of lower-level ones, and such a hierarchy of features is called a deep architecture. Most of these models are based on unsupervised learning of representations." (Wikipedia on "Deep Learning" around March 2012.)

- ***Definition 3****:* "A sub-field of machine learning that is based on learning several levels of representations, corresponding to a hierarchy of features or factors or concepts, where higher-level concepts are defined from lower-level ones, and the same lower-level concepts can help to define many higher-level concepts. Deep learning is part of a broader family of machine learning methods based on learning representations. An observation (e.g., an image) can be represented in many ways (e.g., a vector of pixels), but some representations make it easier to learn tasks of interest (e.g., is this the image of a human face?) from examples, and research in this area attempts to define what makes better representations and how to learn them." see Wikipedia on "Deep Learning" as of this writing in February 2013; see http://en.wikipedia.org/wiki/Deep_learning.

- ***Definition 4****:* "Deep Learning is a new area of Machine Learning research, which has been introduced with the objective of moving Machine Learning closer to one of its original goals: Artificial Intelligence. Deep Learning is about learning multiple levels of representation and abstraction that help to make sense of data such as images, sound, and text." See https://github.com/lisa-lab/DeepLearningTutorials

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Data Science 101 (June 2013)

Deep Learning – A Term To Know

Deep Learning is a new term that is starting to appear in the data science/machine learning news.

- Communications of the ACM just published a story on the topic, Deep Learning Comes of Age.
- Deep Learning was named as one of the Top 10 Breakthrough Technologies of 2013 by MIT Technology Review.
- Jeremy Howard, Chief Scientist at Kaggle declared Deep Learning – The Biggest Data Science Breakthrough of the Decade.
- The New York Times published Scientists See Promise in Deep-Learning Programs

What is Deep Learning?

According to **DeepLearning.net**, the definition goes like this:

"Deep Learning is a new area of Machine Learning research, which has been introduced with the objective of moving Machine Learning closer to one of its original goals: Artificial Intelligence."

Wikipedia provides the following definition:

"Deep learning is set of algorithms in machine learning that attempt to learn layered models of inputs, commonly neural networks. The layers in such models correspond to distinct levels of concepts, where higher-level concepts are defined from lower-level ones, and the same lower-level concepts can help to define many higher-level concepts."

Deep Learning is sometimes referred to as deep neural networks since much of deep learning focuses on artificial neural networks. Artificial neural networks are a technique in computer science modelled after the connections (synapses) of neurons in the brain. Artificial neural networks, sometimes just called neural nets, have been around for about 50 years, but advances in computer processing power and storage are finally allowing neural nets to improve solutions for complex problems such as speech recognition, computer vision, and Natural Language Processing (NLP).

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Useful Sites on Deep Learning

- http://www.cs.toronto.edu/~hinton/
- http://ufldl.stanford.edu/wiki/index.php/UFLDL_Recommended_Readings
- http://ufldl.stanford.edu/wiki/index.php/UFLDL_Tutorial (Andrew Ng's group)
- http://deeplearning.net/reading-list/ (Bengio's group)
- http://deeplearning.net/tutorial/
- **http://deeplearning.net/deep-learning-research-groups-and-labs/**

- Google+ Deep Learning community

http://deeplearning.net/deep-learning-research-groups-and-labs/

Deep Learning Research Groups

Some labs and research groups that are actively working on deep learning:
University of Toronto - **Machine Learning Group** (**Geoff Hinton**, Rich Zemel, Ruslan Salakhutdinov, Brendan Frey, Radford Neal)
Université de Montréal - **Lisa Lab** (**Yoshua Bengio**, Pascal Vincent, Aaron Courville, Roland Memisevic)
New York University – **Yann Lecun**'s and **Rob Fergus**' group
Stanford University – **Andrew Ng**'s group
UBC – **Nando de Freitas**'s group
Google Research – Jeff Dean, Samy Bengio, Jason Weston, Marc'Aurelio Ranzato, Dumitru Erhan, Quoc Le et al
Microsoft Research – **Li Deng** et al
SUPSI – **IDSIA** (Schmidhuber's group)
UC Berkeley – **Bruno Olshausen**'s group
University of Washington – **Pedro Domingos**' group
IDIAP Research Institute - **Ronan Collobert**'s group
University of California Merced – **Miguel A. Carreira-Perpinan**'s group
University of Helsinki - **Aapo Hyvärinen**'s Neuroinformatics group
Université de Sherbrooke – **Hugo Larochelle**'s group
University of Guelph – **Graham Taylor**'s group
University of Michigan – **Honglak Lee**'s group
Technical University of Berlin – **Klaus-Robert Muller**'s group
Baidu – **Kai Yu**'s group
Aalto University – **Juha Karhunen**'s group
U. Amsterdam – **Max Welling**'s group
U. California Irvine – **Pierre Baldi**'s group
Ghent University – **Benjamin Shrauwen**'s group
University of Tennessee – **Itamar Arel**'s group
IBM Research – **Brian Kingsbury** et al
University of Bonn – **Sven Behnke's** group
Gatsby Unit @ University College London – Maneesh Sahani, Yee-Whye Teh, Peter Dayan

Last modified on April 10, 2013, at 1:27 pm by Caglar Gulcehre

MIT Technology Review

10 BREAKTHROUGH TECHNOLOGIES 2013

Deep Learning

With massive amounts of computational power, machines can now recognize objects and translate speech in real time. Artificial intelligence is finally getting smart.

Temporary Social Media

Messages that quickly self-destruct could enhance the privacy of online communications and make people freer to be spontaneous.

Prenatal DNA Sequencing

Reading the DNA of fetuses will be the next frontier of the genomic revolution. But do you really want to know about the genetic problems or musical aptitude of your unborn child?

Additive Manufacturing

Skeptical about 3-D printing? GE, the world's largest manufacturer, is on the verge of using the technology to make jet parts.

Baxter: The Blue-Collar Robot

Rodney Brooks's newest creation is easy to interact with, but the complex innovations behind the robot show just how hard it is to get along with people.

Memory Implants

A maverick neuroscientist believes he has deciphered the code by which the brain forms long-term memories. Next: testing a prosthetic implant for people suffering from long-term memory loss.

Smart Watches

The designers of the Pebble watch realized that a mobile phone is more useful if you don't have to take it out of your pocket.

Ultra-Efficient Solar Power

Doubling the efficiency of a solar cell would completely change the economics of renewable energy. Nanotechnology just might make it possible.

Big Data from Cheap Phones

Collecting and analyzing information from simple cell phones can provide surprising insights into how people move about and behave – and even help us understand the spread of diseases.

Supergrids

A new high-power circuit breaker could finally make highly efficient DC power grids practical.

ICASSP 2013
Vancouver Convention & Exhibition Centre
May 26 - 31, 2013 • Vancouver, Canada

IEEE
IEEE Signal Processing Society

Plenary Keynote (9:50-10:40am, May 28)

Recent Developments in Deep Neural Networks

Geoffrey E. Hinton

UNIVERSITY OF TORONTO

Google

Host: Li Deng

Geoff Hinton

The New York Times

Scientists See Promise in Deep-Learning Programs
John Markoff
November 23, 2012

Rich Rashid in **Tianjin**, October, 25, 2012

The New York Times

Learning Curve: No Longer Just A Human Trait

By JOHN MARKOFF

Using an artificial intelligence technique inspired by theories about how the brain recognizes patterns, technology companies are reporting startling gains in fields as diverse as computer vision, speech recognition and the identification of promising new molecules for designing drugs.

The advances have led to widespread enthusiasm among researchers who design software to perform human activities like seeing, listening and thinking. They offer the promise of machines that converse with humans and perform tasks like driving cars and working in factories, raising the specter of automated robots that could replace human workers.

The technology, called deep learning, has already been put to use in services like Apple's Siri virtual personal assistant, which is based on Nuance Communications' speech recognition service, and in Google's Street View, which uses machine vision to identify specific addresses.

But what is new in recent months is the growing speed and accuracy of deep-learning programs, often called artificial neural networks or just "neural nets" for their resemblance to the neural connections in the brain.

"There has been a number of stunning new results with deep-learning methods," said Yann LeCun, a computer scientist at New York University who did

HAO ZHANG

A voice recognition program translated a speech given by Richard F. Rashid, Microsoft's top scientist, into Mandarin Chinese.

Scientists See Promise in Deep-Learning Programs

From Page A1

seeing in the accuracy of these systems is very rare indeed."

Artificial intelligence researchers are acutely aware of the dangers of being overly optimistic. Their field has long been plagued by outbursts of misplaced enthusiasm followed by equally striking declines.

In the 1960s, some computer scientists believed that a workable artificial intelligence system was just 10 years away. In the 1980s, a wave of commercial start-ups collapsed, leading to what some people called the "A.I. winter."

But recent achievements have impressed a wide spectrum of computer experts. In October, for example, a team of graduate students studying with the University of Toronto computer scientist Geoffrey E. Hinton won the top prize in a contest sponsored by Merck to design software to help find molecules that might lead to new drugs.

From a data set describing the chemical structure of 15 different molecules, they used deep-learning software to determine which molecule was most likely to be an effective drug agent.

The achievement was particularly impressive because the team decided to enter the contest at the last minute and designed its software with no specific knowledge about how the molecules bind to their targets. The students were also working with a relatively small set of data; neural nets typically perform well only with very large ones.

"This is a really breathtaking result because it is the first time that deep learning won, and more significantly it won on a data set that it wouldn't have been ex-

that organizes data science competitions, including the Merck contest.

Advances in pattern recognition hold implications not just for drug development but for an array of applications, including marketing and law enforcement. With greater accuracy, for example, marketers can comb large databases of consumer behavior to get more precise information on buying habits. And improvements in facial recognition are likely to make surveillance technology cheaper and more commonplace.

Artificial neural networks, an idea going back to the 1950s, seek to mimic the way the brain absorbs information and learns from it. In recent decades, Dr. Hinton, 64 (a great-great-grandson of the 19th-century mathematician George Boole, whose work in logic is the foundation for modern digital computers), has pioneered powerful new techniques for helping the artificial networks recognize patterns.

Modern artificial neural networks are composed of an array of software components, divided into inputs, hidden layers and outputs. The arrays can be "trained" by repeated exposures to recognize patterns like images or sounds.

These techniques, aided by the growing speed and power of modern computers, have led to rapid improvements in speech recognition, drug discovery and computer vision.

Deep-learning systems have recently outperformed humans in certain limited recognition tests.

Last year, for example, a program created by scientists at the Swiss A.I. Lab at the University of Lugano won a pattern recogni-

KEITH PENNER

A student team led by the computer scientist Geoffrey E. Hinton used deep-learning technology to design software.

An advance in a technology that can best human brains.

the images in a set of 50,000; the top score in a group of 32 human participants was 99.22 percent, and the average for the humans was 98.84 percent.

This summer, Jeff Dean, a Google technical fellow, and Andrew Y. Ng, a Stanford computer scientist, programmed a cluster of 16,000 computers to train itself to automatically recognize images in a library of 14 million pictures of 20,000 different objects. Although the accuracy rate was low — 15.8 percent — the system did 70 percent better than the most advanced previous one.

Deep learning was given a particularly audacious display at a

lish on a large screen above his head.

Then, in a demonstration that led to stunned applause, he paused after each sentence and the words were translated into Mandarin Chinese characters, accompanied by a simulation of his own voice in that language, which Dr. Rashid has never spoken.

The feat was made possible, in part, by deep-learning techniques that have spurred improvements in the accuracy of speech recognition.

Dr. Rashid, who oversees Microsoft's worldwide research organization, acknowledged that while his company's new speech recognition software made 30 percent fewer errors than previous models, it was "still far from perfect."

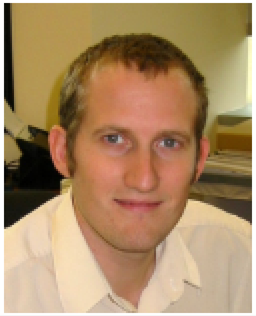
"Rather than having one word in four or five incorrect, now the error rate is one word in seven or eight," he wrote on Microsoft's Web site. Still, he added that this was "the most dramatic change in accuracy" since 1979, "and as we add more data to the training we believe that we will get even better results."

One of the most striking aspects of the research led by Dr. Hinton is that it has taken place largely without the patent restrictions and bitter infighting over intellectual property that characterize high-technology fields.

"We decided early on not to make money out of this, but just to sort of spread it to infect everybody," he said. "These companies are terribly pleased with this."

Referring to the rapid deep-learning advances made possible by greater computing power, and especially the rise of graphics processors, he added:

Keynote Speaker, Vincent Vanhoucke



ICML | Atlanta

International Conference on Machine Learning

16-21 JUNE 2013 ATLANTA

Google

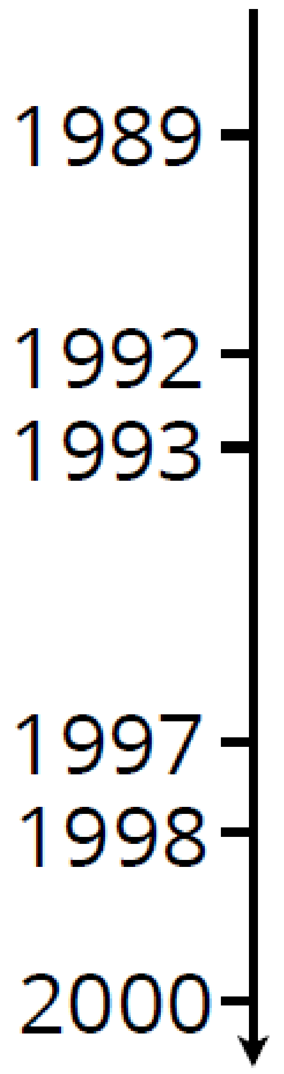
Acoustic Modeling and Deep Learning

June 19th, 2013

Vincent Vanhoucke

Thanks to Vincent for the permission of using his slides & discussions/corrections of information in some slides

Neural Networks for Speech in the 90's



- Time-Delay Neural Networks

  Alex Waibel, Toshiyuki Hanazawa, Geoffrey Hinton, Kiyohiro Shikano, and Kevin J. Lang. "Phoneme recognition using time-delay neural networks. " IEEE Transactions on Acoustics, Speech and Signal Processing, 37, no. 3 (1989): 328-339.

- Recurrent Neural Networks

  Tony Robinson. "A real-time recurrent error propagation network word recognition system", ICASSP 1992.

- Hybrid Systems

  Nelson Morgan, Herve Bourlard, Steve Renals, Michael Cohen, and Horacio Franco. "Hybrid neural network/ hidden Markov model systems for continuous speech recognition." International journal of pattern recognition and artificial intelligence 7, no. 04 (1993): 899-916.

- Bidirectional Recurrent Neural Networks

  Mike Schuster, and Kuldip K. Paliwal. "Bidirectional recurrent neural networks." IEEE Transactions on Signal Processing, 45, no. 11 (1997): 2673-2681.

- Hierarchical Neural Networks

  Jürgen Fritsch and Michael Finke. "ACID/HNN: Clustering hierarchies of neural networks for context-dependent connectionist acoustic modeling." ICASSP 1998.

- TANDEM

  Hynek Hermansky, Daniel PW Ellis, and Sangita Sharma. "Tandem connectionist feature extraction for conventional HMM systems." ICASSP 2000.

Speech Recognition

Speech Recognition + Deep Neural Networks?

Speech Recognition + Deep Neural Networks!

3 months - 10%

word error rate
relative reduction
Voice Search

Application Of Pretrained Deep Neural Networks To Large Vocabulary Speech Recognition, Navdeep Jaitly, Patrick Nguyen, Andrew Senior, Vincent Vanhoucke, Interspeech 2012.

Similar Stories across the Industry

Microsoft

Li Deng
Frank Seide
Dong Yu

IBM

Tara Sainath
Brian Kingsbury

Google

Andrew Senior
Georg Heigold
Marc'Aurelio Ranzato

University of Toronto

Geoff Hinton
George Dahl
Abdel-rahman Mohamed

And many others...

Deep Neural Networks for Acoustic Modeling in Speech Recognition, Geoffrey Hinton, Li Deng, Dong Yu, George Dahl, Abdel-rahman Mohamed, Navdeep Jaitly, Andrew Senior, Vincent Vanhoucke, Patrick Nguyen, Tara Sainath, Brian Kingsbury, IEEE Signal Processing Magazine, Vol. 29, No. 6, November, 2012.

Some of Microsoft's Stories…, Since 2009…

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DL Took off in Speech Recognition from MSR

- Speech recognition: the first big (and real-world) success of deep learning
- From MSR (initial collaboration with Hinton et al., 2009-2010) and then to the entire speech industry
- Got out of “local optimum” of GMM-HMM stayed for many years
- Now used by Microsoft, Google, Apple/Nuance/IBM, Baidu, IFlyTech, etc. doing voice search in the cloud for smart phones (plus many other applications.)

Renaissance of Neural Network --- “Deep Learning,” 2006

-Geoff Hinton invented Deep Belief Networks (DBN) to make neural net learning fast and effective; ***Science, 2006***

- Pre-train each layer from bottom up
- Each pair of layers is an Restricted Boltzmann Machine (RBM)
- Jointly fine-tune all layers using back-propagation

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Industry Scale Deep Learning

Started at MSR, 2009

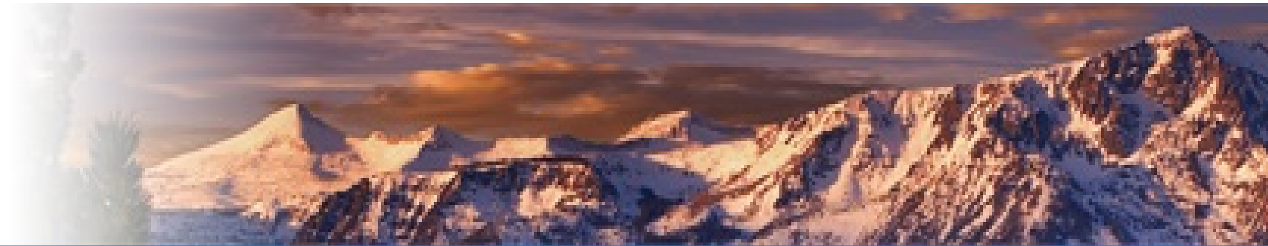
-2008 NIPS: Geoff Hinton & Li Deng reconnected

-Earlier 2009: Initial exploration of DBN/DNN at MSR (image and speech)

-Later 2009: Proof of concept by Mohamed et al.; MSR & Hinton collaborated on applying DBN-DNN to speech feature coding (on spectrogram) and speech recognition

-Dec 2009: NIPS workshop (organizers: Deng, Yu, & Hinton)

Neural Information Processing Systems Foundation



Li Deng, Dong Yu, Geoffrey Hinton

Microsoft Research; Microsoft Research; University of Toronto

Deep Learning for Speech Recognition and Related Applications

7:30am - 6:30pm Saturday, December 12, 2009

Location: Hilton: Cheakamus

Abstract: Over the past 25 years or so, speech recognition technology has b
dominated by a "shallow" architecture --- hidden Markov models (HMMs). Sig
technological success has been achieved using complex and carefully engin
of HMMs. The next generation of the technology requires solutions to remain
challenges under diversified deployment environments. These challenges, no
addressed in the past, arise from the many types of variability present in the
generation process. Overcoming these challenges is likely to require "deep" a
with efficient learning algorithms. For speech recognition and related sequen
recognition applications, some attempts have been made in the past to deve
computational architectures that are "deeper" than conventional HMMs, such

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Industry Scale Deep Learning

Continued at MSR, 2010, 2011…

-2010: slowly more people in MSR-speech joined DBN-DNN research

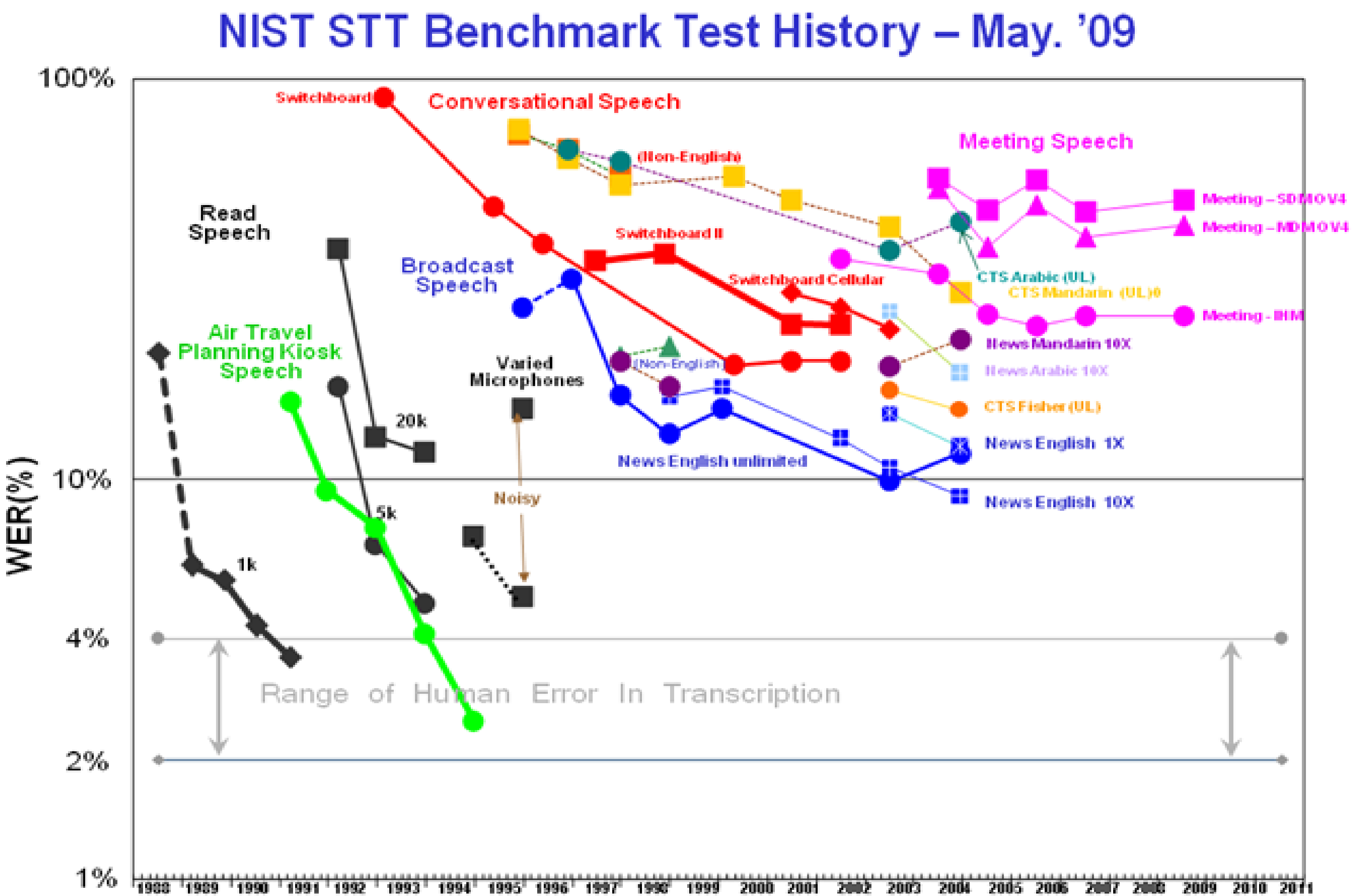
-July 2010: success of bottleneck feature coding using speech spectrogram; Interspeech-2010 paper Deng/Hinton et al.

-**August 2010**: success of DNN in large-vocabulary speech recognition (**voice search**); paper in ICASSP-2011 (Dahl/Yu/Deng)

-Oct 2010: MSR/MSRA collaboration started on Switchboard task

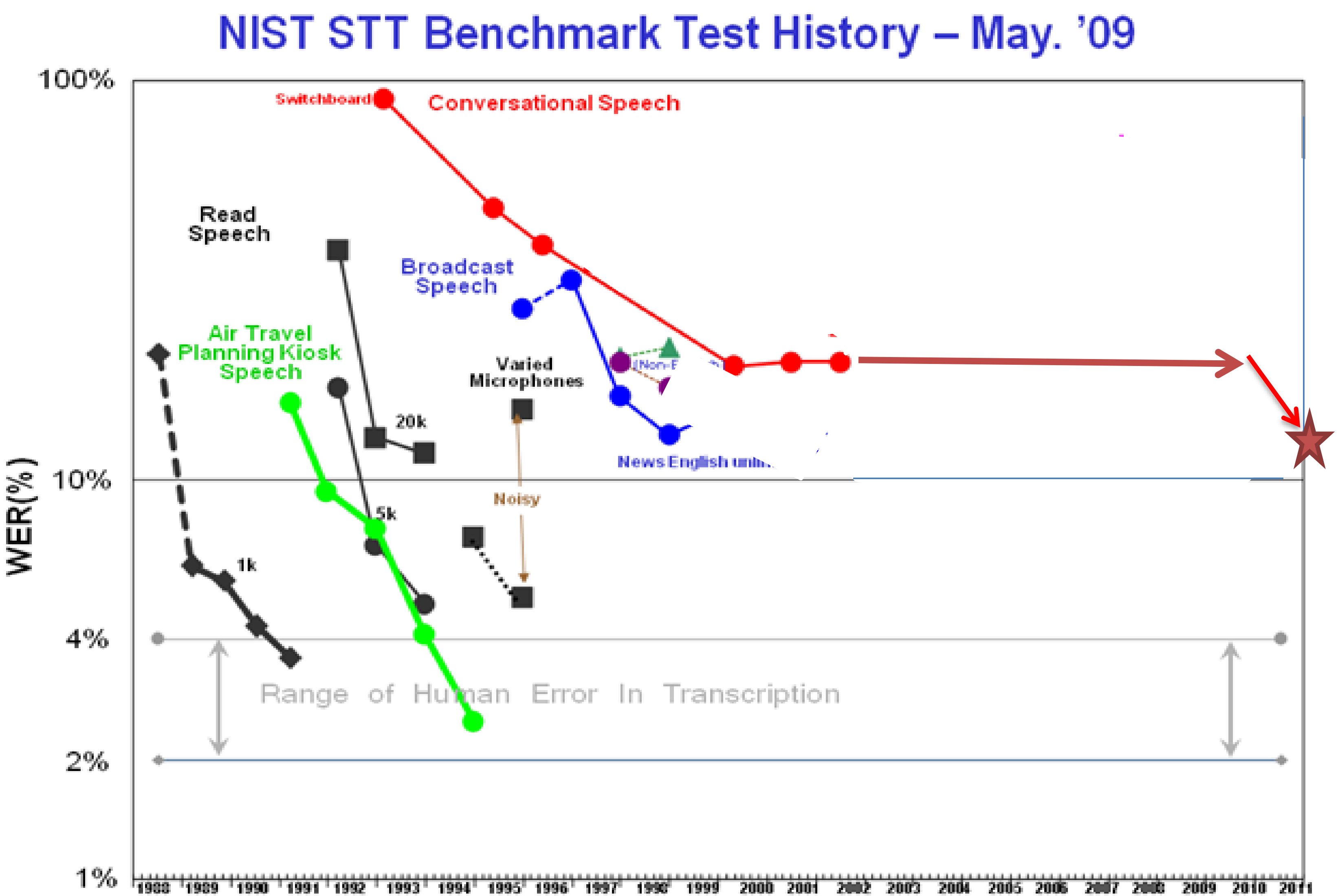
-March 2011: Success in the Switchboard task by MSR/MSRA; Interspeech-2011: Seide/Yu, et al. Success of **deep stacking net**: Deng/Yu/Platt.

The History of
Automatic Speech Recognition Evaluations at NIST



1989 1999 2011

The History of
Automatic Speech Recognition Evaluations at NIST



1989 2011

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IEEE TRANSACTIONS ON AUDIO, SPEECH, AND LANGUAGE PROCESSING, VOL. 21, NO. 5, MAY 2013 pp. 1060-1089 1

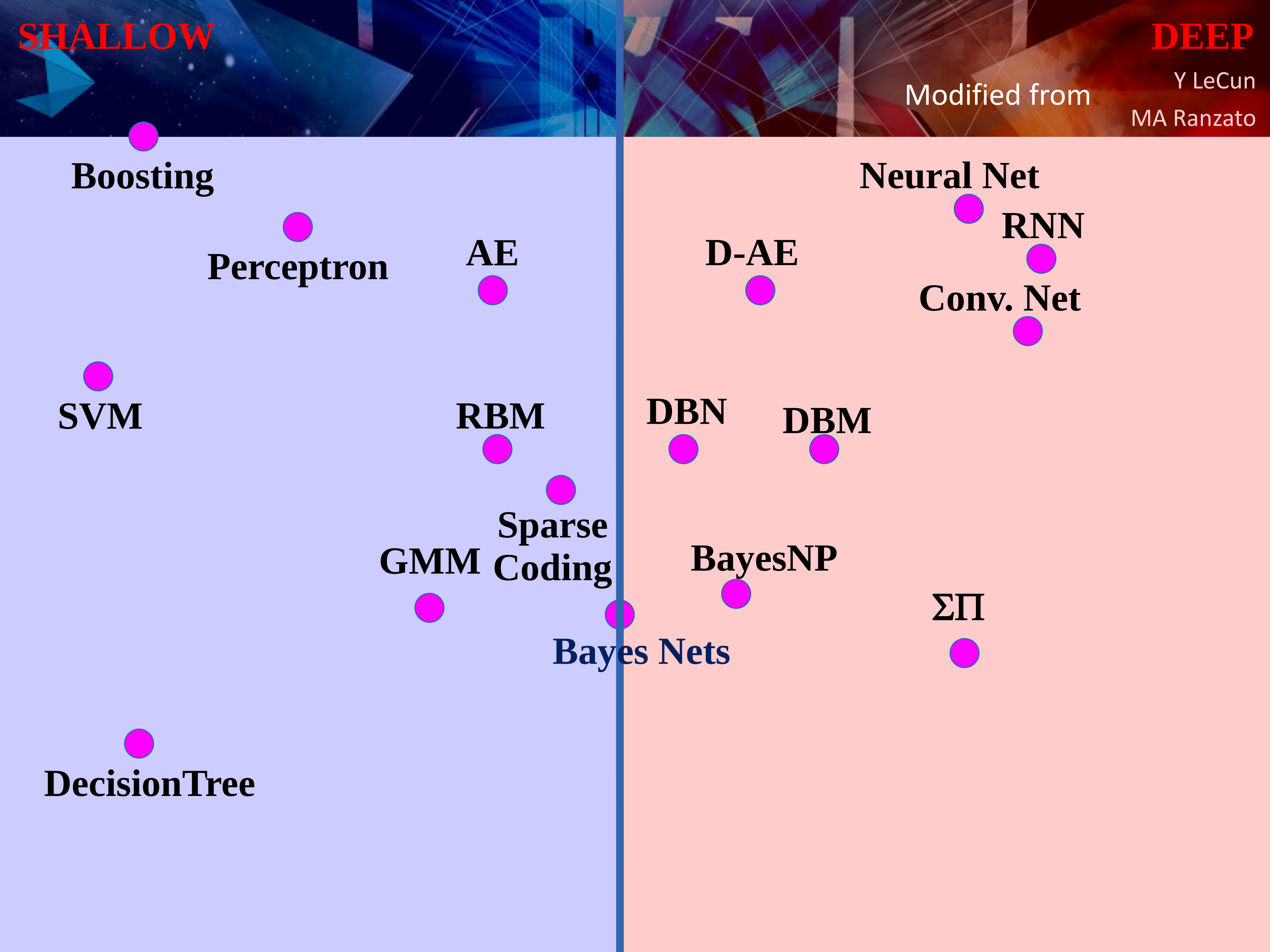
Machine Learning Paradigms for Speech Recognition: An Overview

Li Deng, *Fellow, IEEE*, and Xiao Li, *Member, IEEE*

***Abstract*—Automatic Speech Recognition (ASR) has historically been a driving force behind many machine learning (ML) techniques, including the ubiquitously used hidden Markov model, discriminative learning, structured sequence learning, Bayesian learning, and adaptive learning. Moreover, ML can and occasionally does use ASR as a large-scale, realistic application to rigorously test the effectiveness of a given technique, and to inspire new problems arising from the inherently sequential and dynamic nature of speech. On the other hand, even though ASR is available commercially for some applications, it is largely an unsolved problem—for almost all applications, the performance of ASR is not on par with human performance. New insight from modern ML methodology shows great promise to advance the state-of-the-art in ASR technology. This overview article provides readers with an overview of modern ML techniques as utilized in the current and as relevant to future ASR research and systems. The intent is to foster further cross-pollination between the ML**

community to make assumptions about a problem, develop precise mathematical theories and algorithms to tackle the problem given those assumptions, but then evaluate on data sets that are relatively small and sometimes synthetic. ASR research, on the other hand, has been driven largely by rigorous empirical evaluations conducted on very large, standard corpora from real world. ASR researchers often found formal theoretical results and mathematical guarantees from ML of less use in preliminary work. Hence they tend to pay less attention to these results than perhaps they should, possibly missing insight and guidance provided by the ML theories and formal frameworks even if the complex ASR tasks are often beyond the current state-of-the-art in ML.

This overview article is intended to provide readers of IEEE TRANSACTIONS ON AUDIO, SPEECH, AND LANGUAGE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Y LeCun
MA Ranzato
Boosting
Neural Net
RNN
Perceptron
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D-AE
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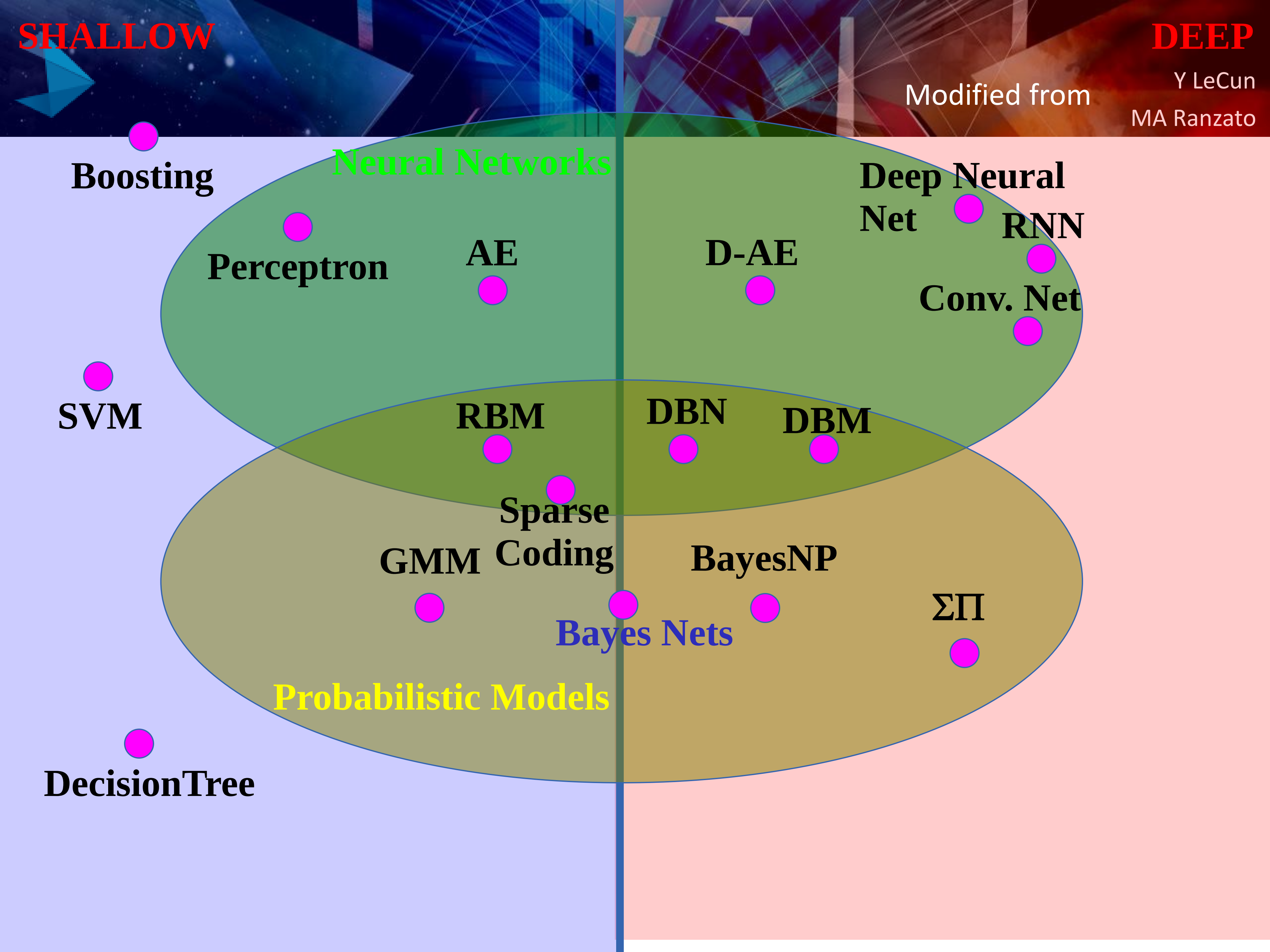
Sparse
Coding
GMM
BayesNP
ΣΠ
Bayes Nets
DecisionTree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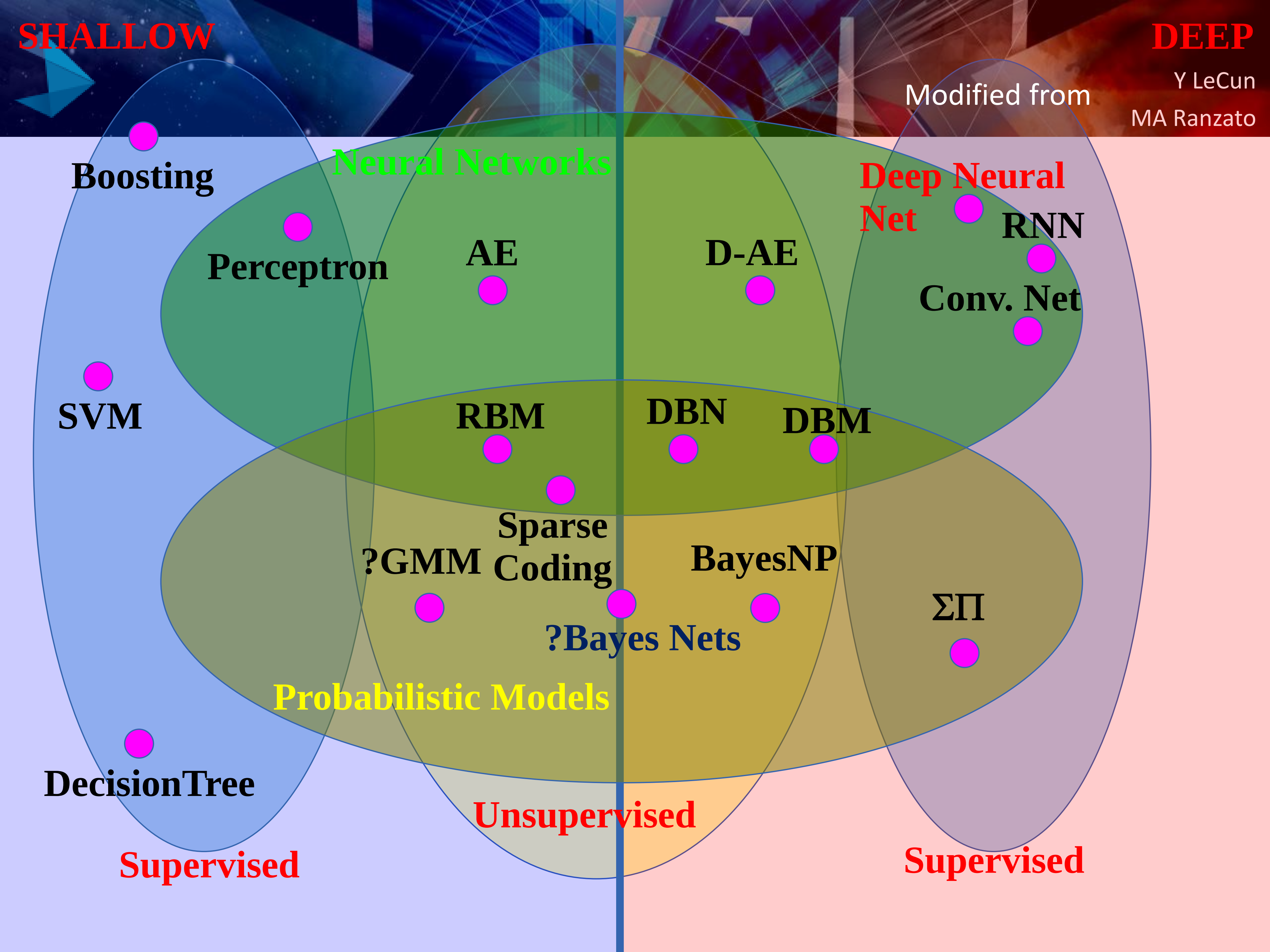
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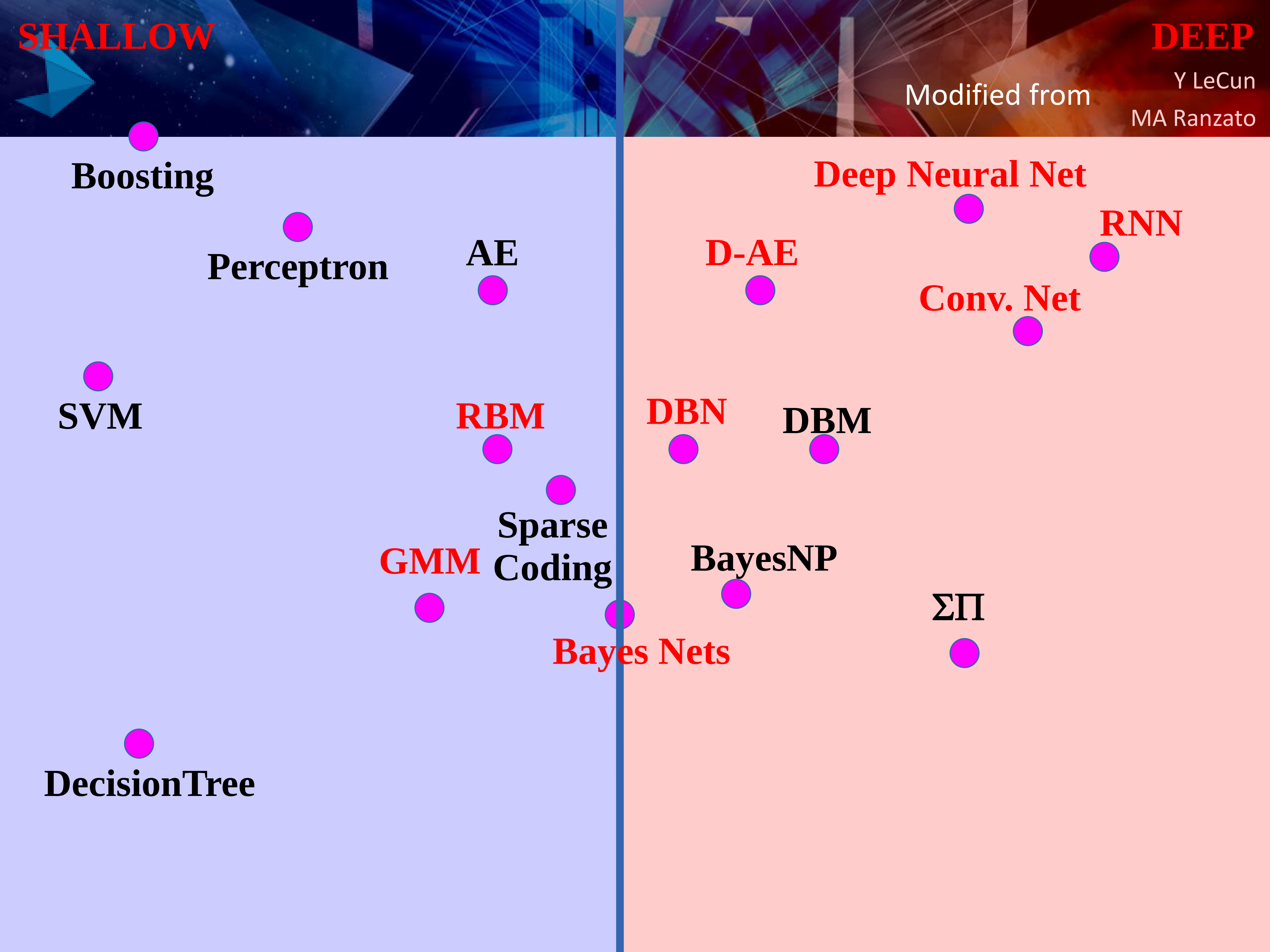
?Bayes Nets
Probabilistic Models
DecisionTree
Unsupervised
Supervised
Supervised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PART II: Deeper Substance of DL

---**Technical introduction: RBM, DBN, DNN, CNN, RNN**

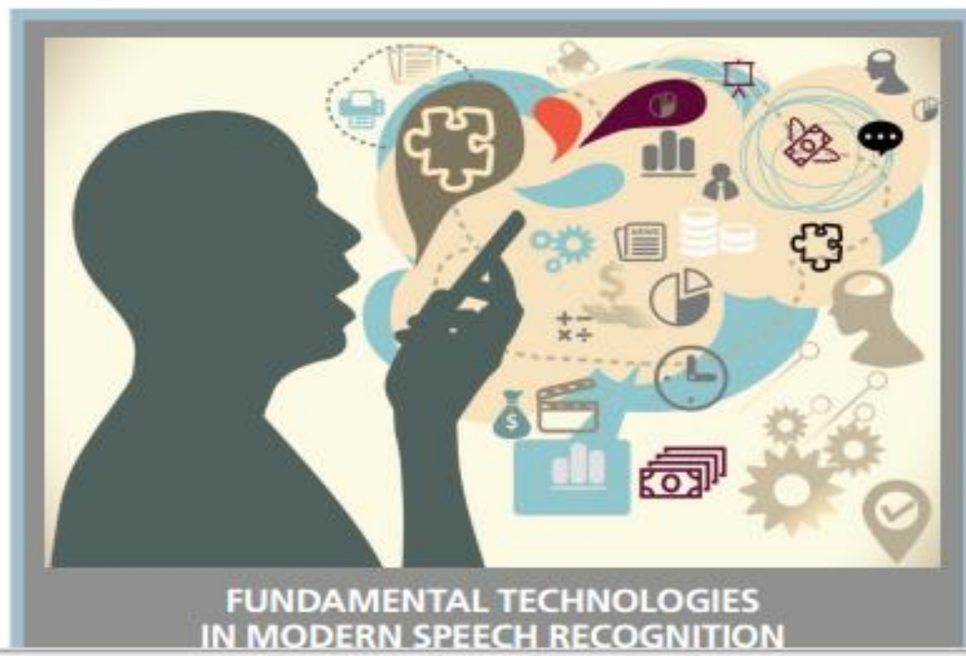
---Advanced: 2 examples of incorporating domain knowledge (speech) into DL architectures

---Novel DL architectures and recent experiments

Geoffrey Hinton, Li Deng, Dong Yu, George E. Dahl, Abdel-rahman Mohamed, Navdeep Jaitly, Andrew Senior, Vincent Vanhoucke, Patrick Nguyen, Tara N. Sainath, and Brian Kingsbury

Deep Neural Networks for Acoustic Modeling in Speech Recognition

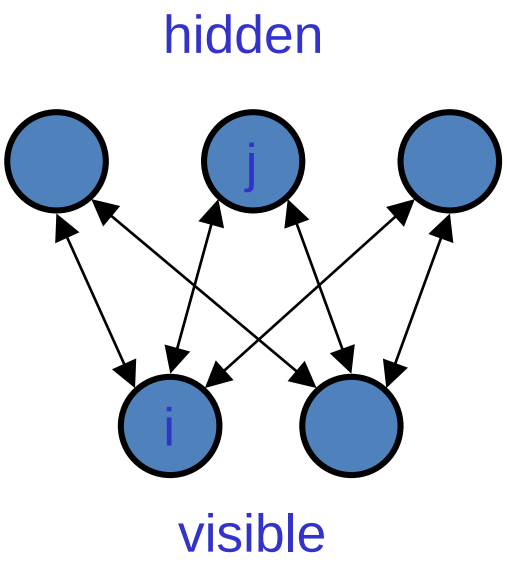
[The shared views of four research groups]



IEEE Signal Proc.
Magazine
November, 2012

Restricted Boltzmann Machines (RBM)

- We restrict the connectivity to make learning easier.
  - Only one layer of hidden units.
  - No connections between hidden units.
- In an RBM, the hidden units are conditionally independent given the visible states.
- So we can quickly get an unbiased sample from the posterior distribution when given a data-vector.



RBM: Weights → Energies → Probabilities

- Joint distribution $\mathrm{p}(\mathbf{v},\mathbf{h};\theta)$ is defined in terms of an energy function $\mathrm{E}(\mathbf{v},\mathbf{h};\theta)$

$$\mathrm{p}(\mathbf{v},\mathbf{h};\theta) = \frac{exp\left(-\mathrm{E}(\mathbf{v},\mathbf{h};\theta)\right)}{Z}$$

- For a Bernoulli-Bernoulli RBM

$$\mathrm{E}(\mathbf{v},\mathbf{h};\theta) = -\sum_{i=1}^{V}\sum_{j=1}^{H} w_{ij}\, v_i h_j - \sum_{i=1}^{V} b_i v_i - \sum_{j=1}^{H} a_j h_j$$

- For a Gaussian-Bernoulli RBM

$$\mathrm{E}(\mathbf{v},\mathbf{h};\theta) = -\sum_{i=1}^{V}\sum_{j=1}^{H} w_{ij}\, v_i h_j + \frac{1}{2}\sum_{i=1}^{V}(v_i - b_i)^2 - \sum_{j=1}^{H} a_j h_j$$

- $\mathrm{p}(\mathbf{v},\mathbf{h};\theta)$ → generative model!

Restricted Boltzmann Machine (RBM)

- Conditional probabilities are very easy to calculate
- For a Bernoulli-Bernoulli RBM

Inference

$$p(h_j = 1|\mathbf{v};\theta) = \sigma\left(\sum_{i=1}^{V} w_{ij}\, v_i + a_j\right)$$

synthesis

$$p(v_i = 1|\mathbf{h};\theta) = \sigma\left(\sum_{j=1}^{H} w_{ij}\, h_j + b_i\right)$$

- For a Gaussian-Bernoulli RBM

Inference

$$p(h_j = 1|\mathbf{v};\theta) = \sigma\left(\sum_{i=1}^{V} w_{ij}\, v_i + a_j\right)$$

$$p(v_i|\mathbf{h};\theta) = N\left(\sum_{j=1}^{H} w_{ij}\, h_j + b_i, 1\right)$$

- Proof next page. (This is a “neural net” with stochastic units rather than the deterministic MLP that you may be more familiar with)

IEEE Trans. Audio, Speech, Lang. Proc., Jan 2012

$$
\begin{aligned}
P(\mathbf{h}|\mathbf{v}) &= \frac{e^{-E(\mathbf{v},\mathbf{h})}}{\sum_{\tilde{\mathbf{h}}} e^{-E(\mathbf{v},\tilde{\mathbf{h}})}} \\
&= \frac{e^{\mathbf{b}^T\mathbf{v}+\mathbf{c}^T\mathbf{h}+\mathbf{v}^T\mathbf{W}\mathbf{h}}}{\sum_{\tilde{\mathbf{h}}} e^{\mathbf{b}^T\mathbf{v}+\mathbf{c}^T\tilde{\mathbf{h}}+\mathbf{v}^T\mathbf{W}\tilde{\mathbf{h}}}} \\
&= \frac{e^{\mathbf{c}^T\mathbf{h}+\mathbf{v}^T\mathbf{W}\mathbf{h}}}{\sum_{\tilde{\mathbf{h}}} e^{\mathbf{c}^T\tilde{\mathbf{h}}+\mathbf{v}^T\mathbf{W}\tilde{\mathbf{h}}}} \\
&= \frac{\prod_i e^{c_i h_i+\mathbf{v}^T\mathbf{W}_{*,i}h_i}}{\sum_{\tilde{h}_1}\cdots\sum_{\tilde{h}_N}\prod_i e^{c_i\tilde{h}_i+\mathbf{v}^T\mathbf{W}_{*,i}\tilde{h}_i}} \\
&= \frac{\prod_i e^{-\gamma_i(\mathbf{v},h_i)}}{\sum_{\tilde{h}_1}\cdots\sum_{\tilde{h}_N}\prod_i e^{-\gamma_i(\mathbf{v},\tilde{h}_i)}} \\
&= \frac{\prod_i e^{-\gamma_i(\mathbf{v},h_i)}}{\prod_i\sum_{\tilde{h}_i} e^{-\gamma_i(\mathbf{v},\tilde{h}_i)}} \\
&= \prod_i \frac{e^{-\gamma_i(\mathbf{v},h_i)}}{\sum_{\tilde{h}_i} e^{-\gamma_i(\mathbf{v},\tilde{h}_i)}} \qquad (5) \\
&= \prod_i P(h_i|\mathbf{v}). \qquad (6)
\end{aligned}
$$

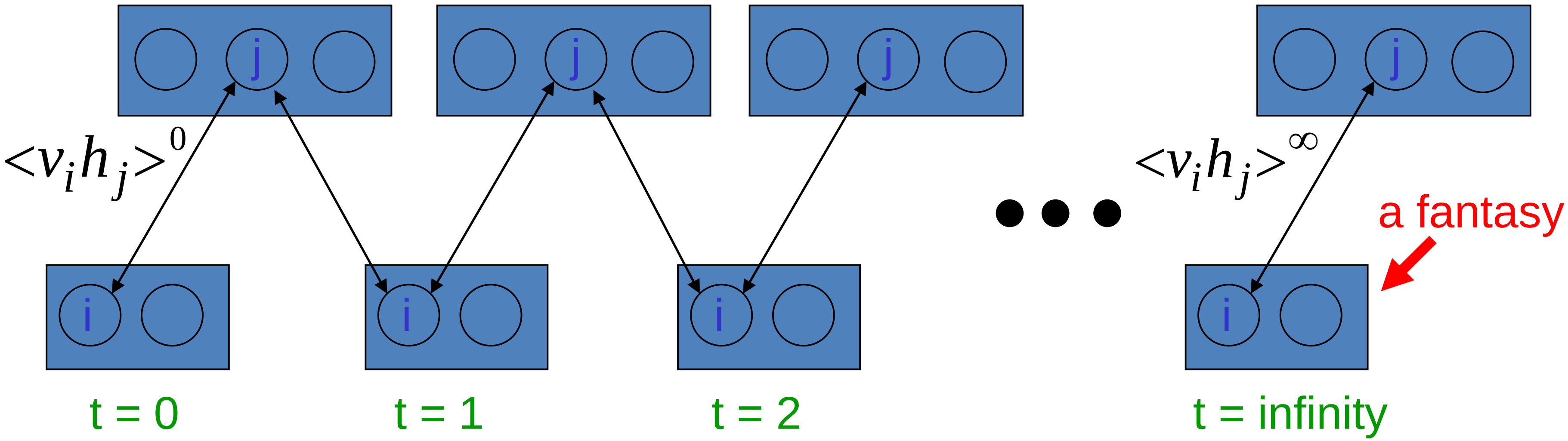
Since the $h_i \in \{0,1\}$, the sum in the denominator of equation (5) has only two terms and thus

$$
\begin{aligned}
P(h_i=1|\mathbf{v}) &= \frac{e^{-\gamma_i(\mathbf{v},1)}}{e^{-\gamma_i(\mathbf{v},1)}+e^{-\gamma_i(\mathbf{v},0)}} \\
&= \sigma(c_i+\mathbf{v}^T\mathbf{W}_{*,i}),
\end{aligned}
$$

yielding

$$P(\mathbf{h}=1|\mathbf{v}) = \sigma(\mathbf{c}+\mathbf{v}^T\mathbf{W}), \qquad (7)$$

Maximum likelihood learning for RBM



Start with a training vector on the visible units.

Then alternate between updating all the hidden units in parallel and updating all the visible units in parallel.

$$\frac{\partial \log p(v)}{\partial w_{ij}} = <v_i h_j>^0 - <v_i h_j>^\infty$$

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Training RBMs

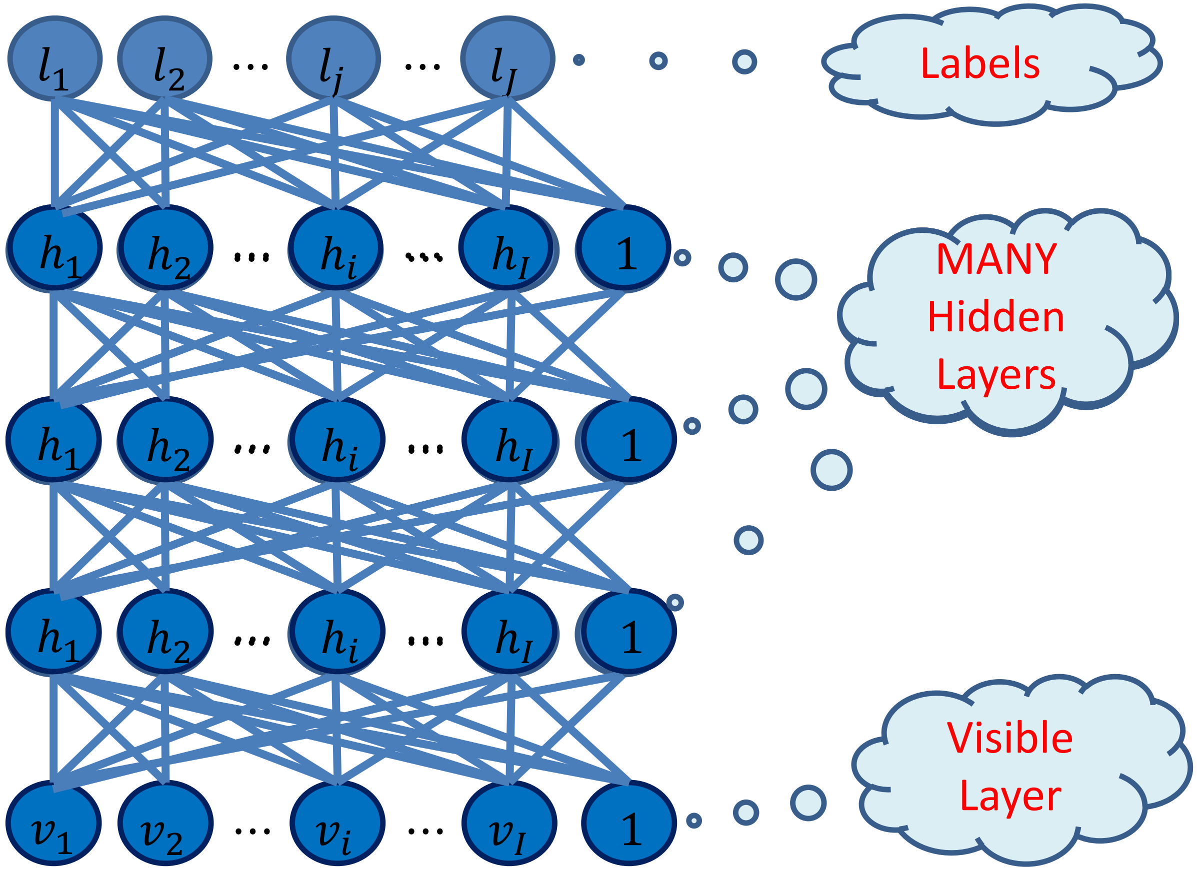
- $\Delta w_{ij} = \langle v_i h_j \rangle_{data} - \langle v_i h_j \rangle_{model}$
- Approximate $\langle v_i h_j \rangle_{model}$
  i. Initialize $\mathbf{v_0}$ at data
  ii. Sample $\mathbf{h_0} \sim \boldsymbol{p}(\mathbf{h}|\mathbf{v_0})$
  iii. Sample $\mathbf{v_1} \sim \boldsymbol{p}(\mathbf{v}|\mathbf{h_0})$
  iv. Sample $\mathbf{h_1} \sim \boldsymbol{p}(\mathbf{h}|\mathbf{v_1})$
  v. Call ($\mathbf{v_1}$, $\mathbf{h_1}$) a sample from the model.
- ($\mathbf{v}_\infty$, $\mathbf{h}_\infty$) is a true sample from the model. ($\mathbf{v_1}$, $\mathbf{h_1}$) is a very rough estimate but worked
- Contrastive divergence algorithm (CD)

Building a Deep Network

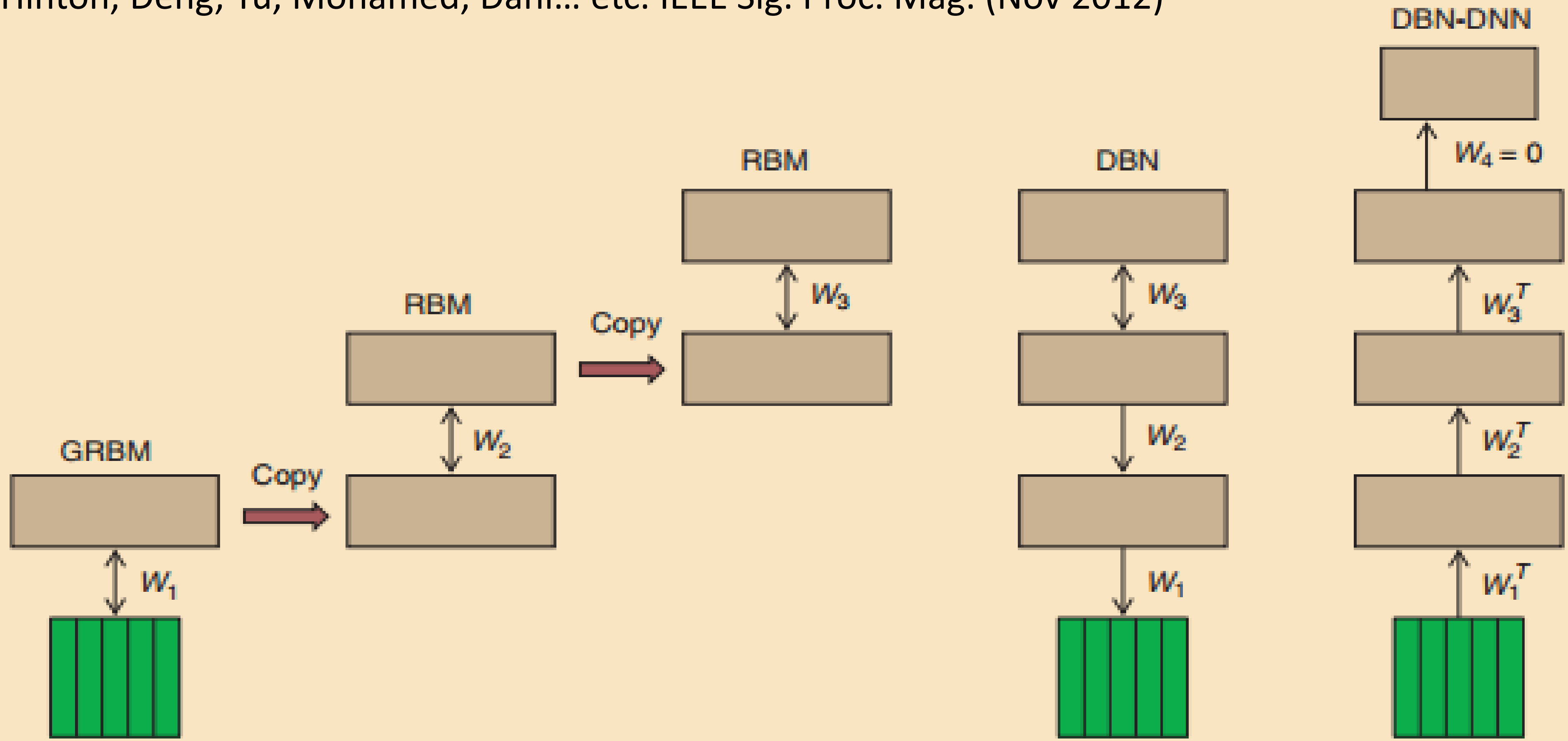
- This is the main reason why RBM's are interesting (as a building block)
- First train a layer of hidden units that receive input directly from the data (image, speech, coded text, etc).
- Then treat the activations of hidden units (the trained "features") as if they were "data" and learn features of features in a second hidden layer.
- It can be proved that each time we add another layer of features we improve a variational lower bound on the log probability of the training data.
  - The proof is complicated (Hinton et al, 2006)
  - Based on an equivalence between an RBM and a deep directed model

Deep Belief Net (DBN) & Deep Neural Net (DNN)

- DBN: Undirected at top two layers which is an RBM; directed Bayes net (top-down) at lower layers (good for **synthesis and recognition**)
- DNN: Multi-layer perceptron (bottom up) + unsupervised pre-training w. RBM weights (good for **recognition only**)



Hinton, Deng, Yu, Mohamed, Dahl... etc. IEEE Sig. Proc. Mag. (Nov 2012)



First train a stack of three models each of which has one hidden layer. Each model in the stack treats the hidden variables of the previous model as data.

Then compose them into a single Deep Belief Network.

Then add outputs and train the DNN with backprop.

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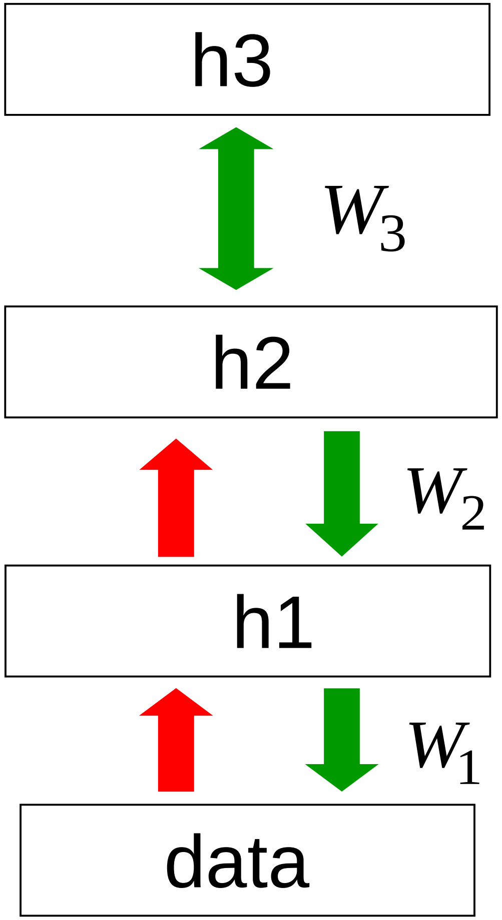
Quiz Questions

1. DBN & DNN: which one is generative? Which one is discriminative?
2. How can a generative model be used for recognition? (Bayes rule as for HMM speech recognition)
3. How does DBN do synthesis?
4. How does DBN do recognition?
5. How does DNN do recognition?
6. For recognition, is RBN or DNN better?
7. What is the difference between DBN and Dynamic Bayes Net (a.k.a. “DBN”)?

The Answer to Quiz Question 3:

- To generate data:
  1. Get an equilibrium sample from the top-level RBM by performing alternating Gibbs sampling for a long time.
  2. Perform a top-down pass to get states for all the other layers.

So the lower level bottom-up connections are not part of the generative model. They are just used for inference.

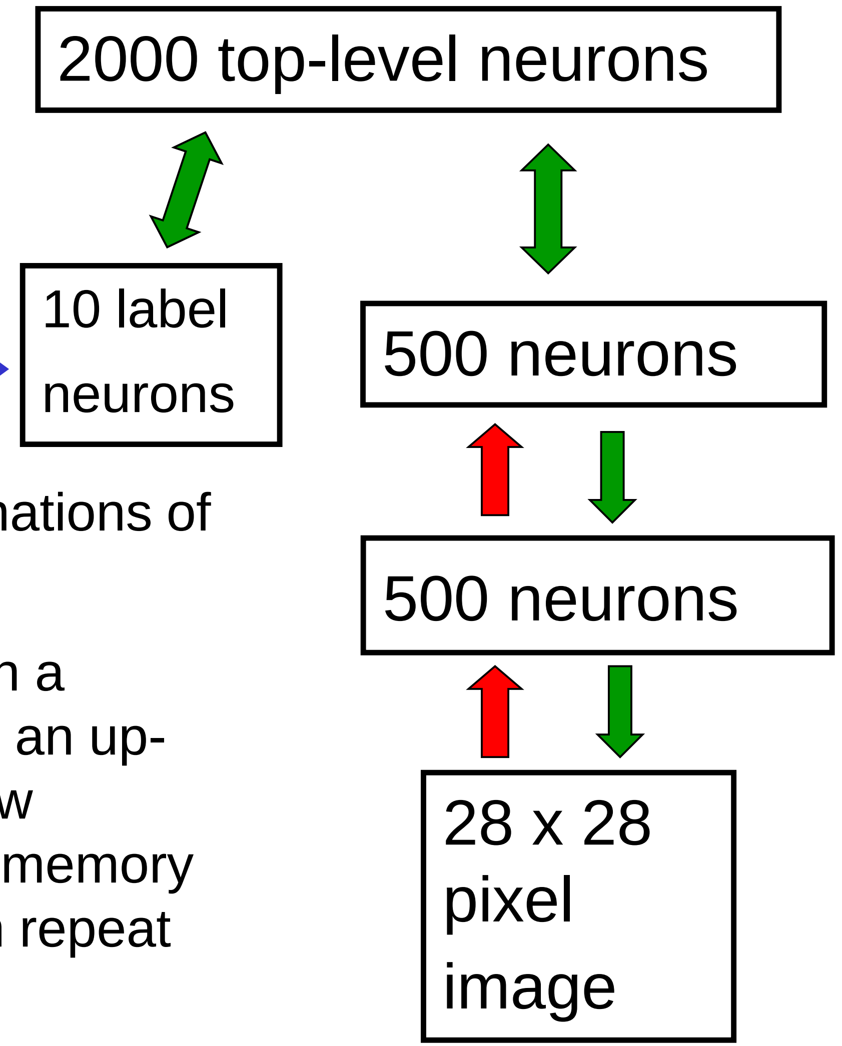


Samples generated by letting the associative memory run with one label clamped. There are 1000 iterations of alternating Gibbs sampling between samples (example from Hinton) .

Answer to Quiz Question 4: Example of digit/image recognition by DBN

The top two layers form an associative memory whose energy landscape models the low dimensional manifolds of the digits

The energy valleys have names →



The model learns to generate combinations of labels and images.

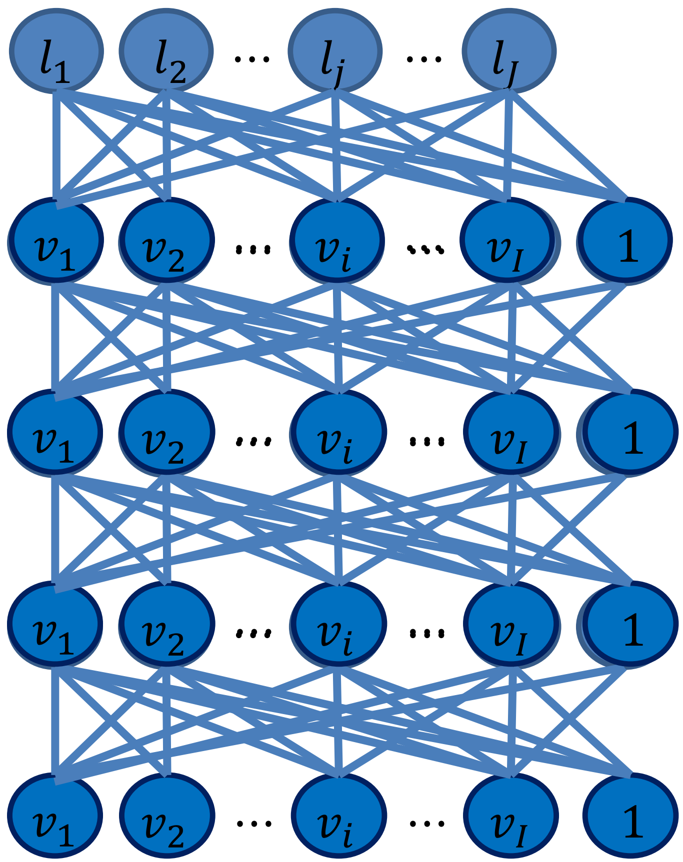
To perform recognition we start with a neutral state of the label units and do an up-pass from the image followed by a few iterations of the top-level associative memory ---> probability of that digit label; then repeat for all digit labels; then compare.

(slide modified from Hinton)

DBN & DNN: Fine-tuning for discrimination

- First learn one layer at a time greedily.
- Then treat this as “pre-training” that finds a good initial set of weights which can be fine-tuned by a local search procedure.
- For DBN: Contrastive wake-sleep (see Hinton’s)
- **For DNN: Back-propagation**
  – This overcomes many of the limitations of standard backpropagation (if you do not have large labeled training data).

DNN with class posteriors (not DBN)



- As stacked RBMs
- Pre-train each layer from bottom up by considering each pair of layers as an RBM.
- Transform the output of the last hidden layer into a multinomial distribution using the softmax operation

$$\mathrm{p}(l = k|\mathbf{h}; \theta) = \frac{exp\left(\sum_{i=1}^{H} \lambda_{ik} h_i + a_k\right)}{Z(\boldsymbol{h})}$$

- Why? Needed for (ASR) sequence recognition (not needed for static or frame-level recognition)
- For ASR: Use GMM-HMM forced alignment to get the label for the final layer when using frame-level training.
- Jointly fine-tune all layers using back-propagation algorithm.

The current wisdom on unsupervised pre-training

- Pre-training achieves two things:
  - It makes optimization easier.
  - It reduces overfitting.
- We now know more about how to initialize weights sensibly by hand.
  - So unsupervised pre-training is not required to make the optimization work.
- Unsupervised pre-training is still very effective at preventing over-fitting when labeled data is scarce.
  - It is not needed when labeled data is abundant.

Hinton: ICASSP-2013

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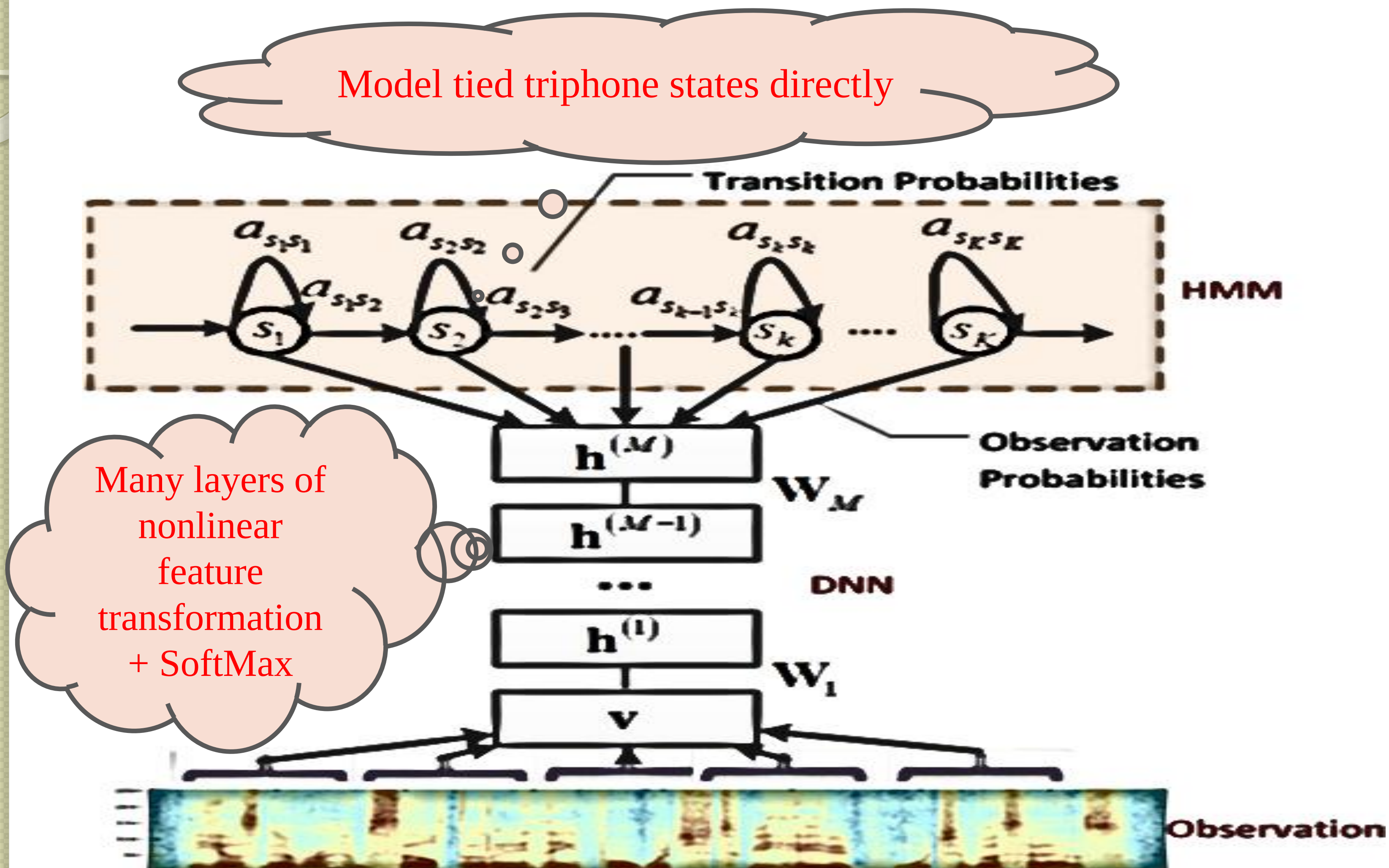
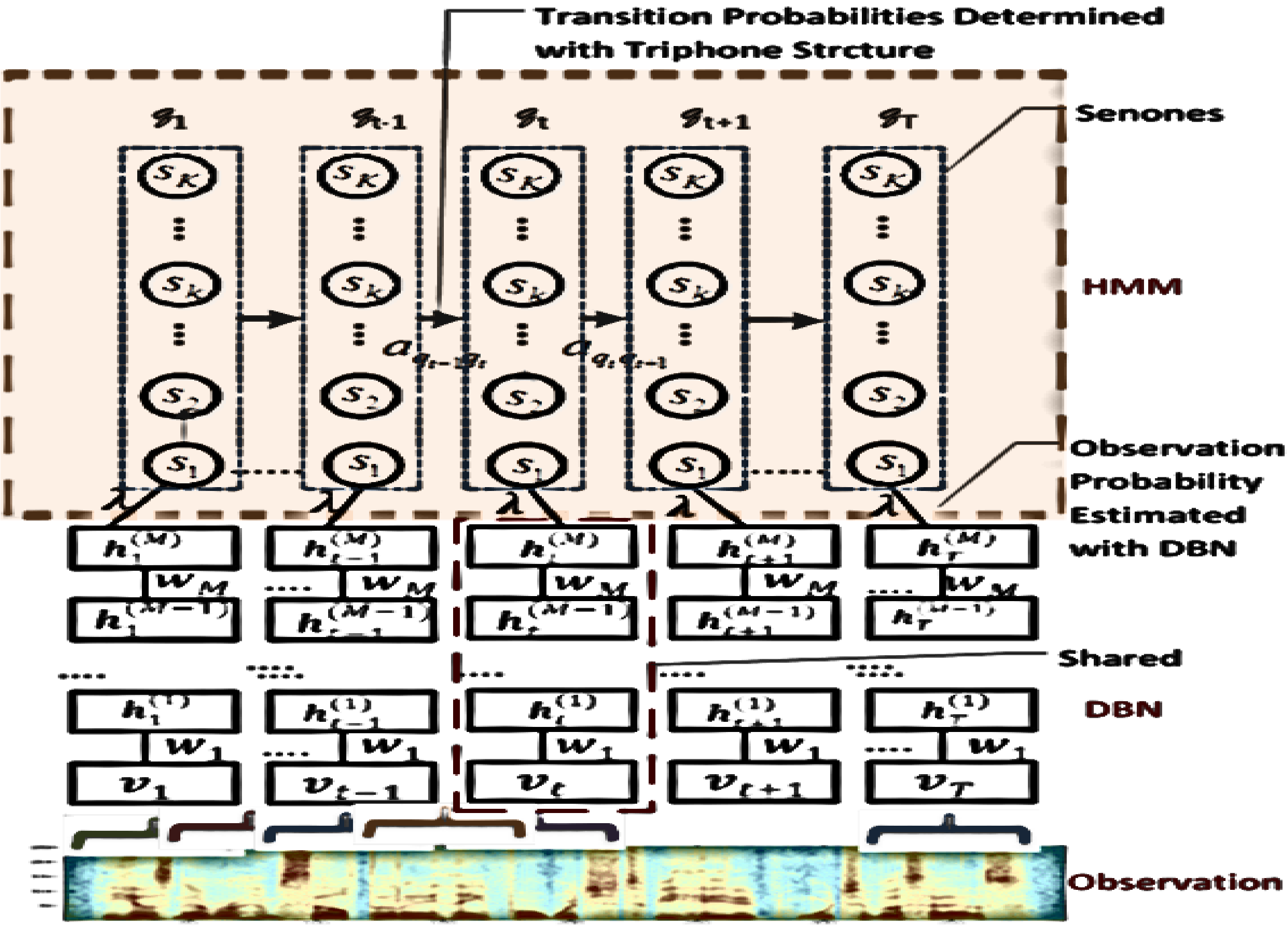
DNN-HMM

(replacing GMM only; longer MFCC/filter-back windows w. no transformation)



CD-DNN-HMM: Architecture



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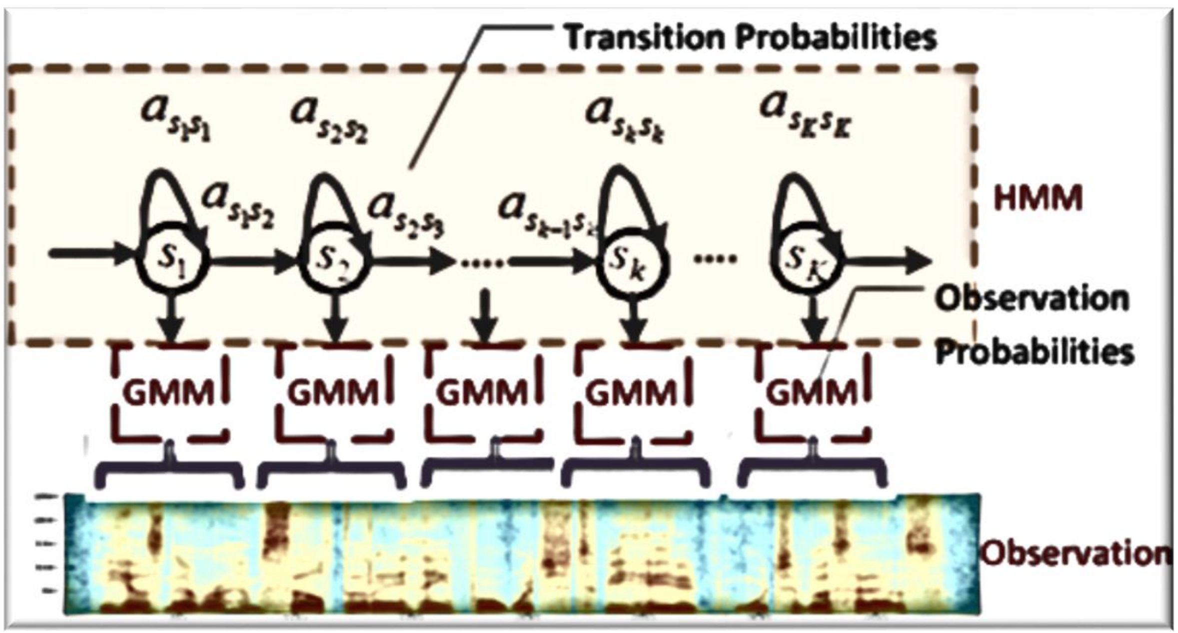
(Shallow) GMM-HMM

- Model frames of acoustic data with two stochastic processes:
  - A hidden Markov process to model state transition
  - A Gaussian mixture model to generate observations
- Train with maximum likelihood criterion using EM followed by discriminative training (e.g. MPE)



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Voice Search with DNN-HMM

- First attempt in using deep models for large vocabulary speech recognition (summer 2010)
- Published in ICASSP-2011 & 2012 Special issue of T-ASLP:

30 IEEE TRANSACTIONS ON AUDIO, SPEECH, AND LANGUAGE PROCESSING, VOL. 20, NO. 1, JANUARY 2012

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

George E. Dahl, Dong Yu, *Senior Member, IEEE*, Li Deng, *Fellow, IEEE*, and Alex Acero, *Fellow, IEEE*

***Abstract*—We propose a novel context-dependent (CD) model for**

fields (CRFs) [18]–[20], hidden CRFs [21], [22], and segmental

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MSR Key Innovations (2009-2013)

- Scale the success to large industrial speech tasks
  - Grew output neurons from context-independent phones (100-200) to context-dependent ones (9k-32k)
  - Motivated initially by saving huge MSFT investment in huge speech decoder software infrastructure (e.g. Entropic acquisition)
  - Extremely fast decoder
  - Developed novel deep learning architectures & techniques: DCN/DSN, tensor-DSN, kernel-DCN, tensor-DNN, etc.
- Engineering for large systems:
  - Expertise in DNN **and** speech recognition
  - Close collaboration among MSRR, MSRA, & speech product teams

(Deng, Yu, Seide, Gang Li, Jinyu Li, Jui-Ting Huang, Yifan Gong, etc.)

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Some Recent News by Reporters

- DNN Research Improves Bing Voice Search (very fast decoder)
- How technology can bridge language gaps: Speech-to-speech translation promises to help connect our world
- Scientists See Promise in Deep-Learning Programs (NYT: speech to speech)
- Microsoft Research shows a promising new breakthrough in speech translation technology
- Bing Makes Voice Recognition on Windows Phone More Accurate and Twice as Fast
- Microsoft revs speedier, smarter speech recognition for phones

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(Including impact and recent history of DL (Deep Neural Net, DNN) in speech recognition)

Outline

PART II: Deeper Substance of DL

---Technical introduction: RBM, DBN, DNN, DNN-HMM, CNN, RNN

---Examples of incorporating domain knowledge (about speech) into DL architectures

1. Hidden/articulatory Speech dynamics into RNN
2. Speech invariance/class-discrim.into deep-CNN

---A few new, promising DL architectures

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Outline

~~PART I: Basics of Deep Learning (DL)~~

(including impact and recent history of DL (Deep Neural Net, DNN) in speech recognition)

PART II: Deeper Substance of DL

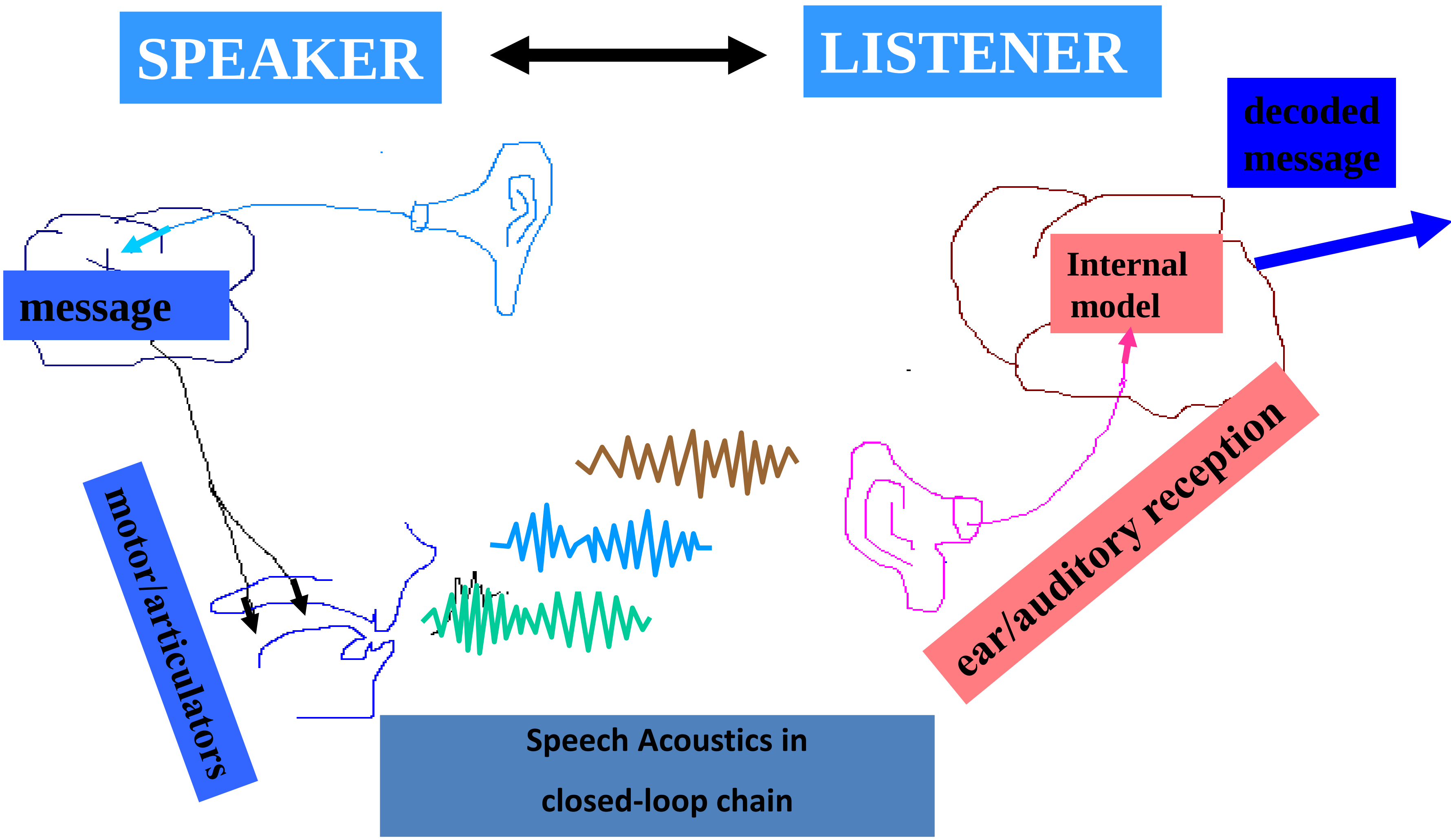
---Example 1: incorporating domain knowledge:

Hidden/Deep Dynamics in Human Speech

Deep/Dynamic Models are Natural for Speech

- **Hierarchical structure in human speech generation**
  - Global concept/semantics formation
  - Word sequence formation / prosodic planning
  - Phonological encoding (phones, distinctive features)
  - Phonetic encoding (motor commands, articulatory targets)
  - Articulatory dynamics
  - Acoustic dynamics (clean speech)
  - Distorted speech
  - Interactions between speakers and listener/machine
- Hierarchical structure in human speech perception
  - Cochlear nonlinear spectral analysis
  - Attribute/phonological-feature detection at higher level(s)
  - Phonemic and syllabic detection at still higher level(s)
  - Word and sequence detection
  - Syntactic analysis and semantic understanding at deeper auditory cortex

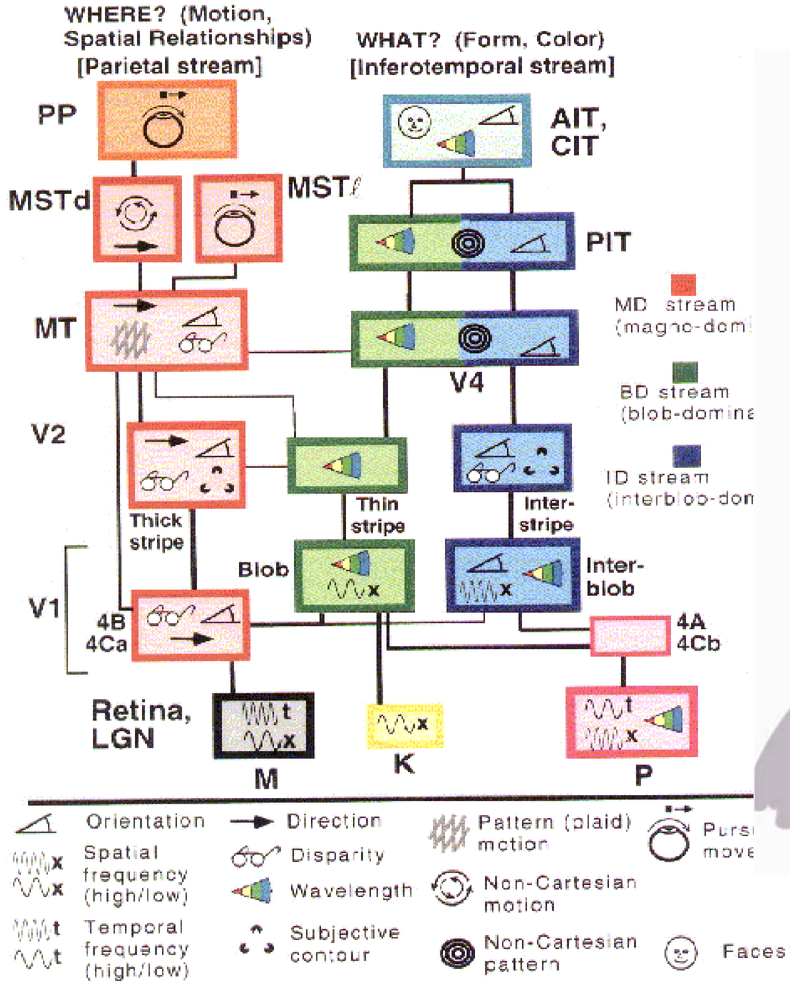
Production & Perception: Closed-Loop Chain

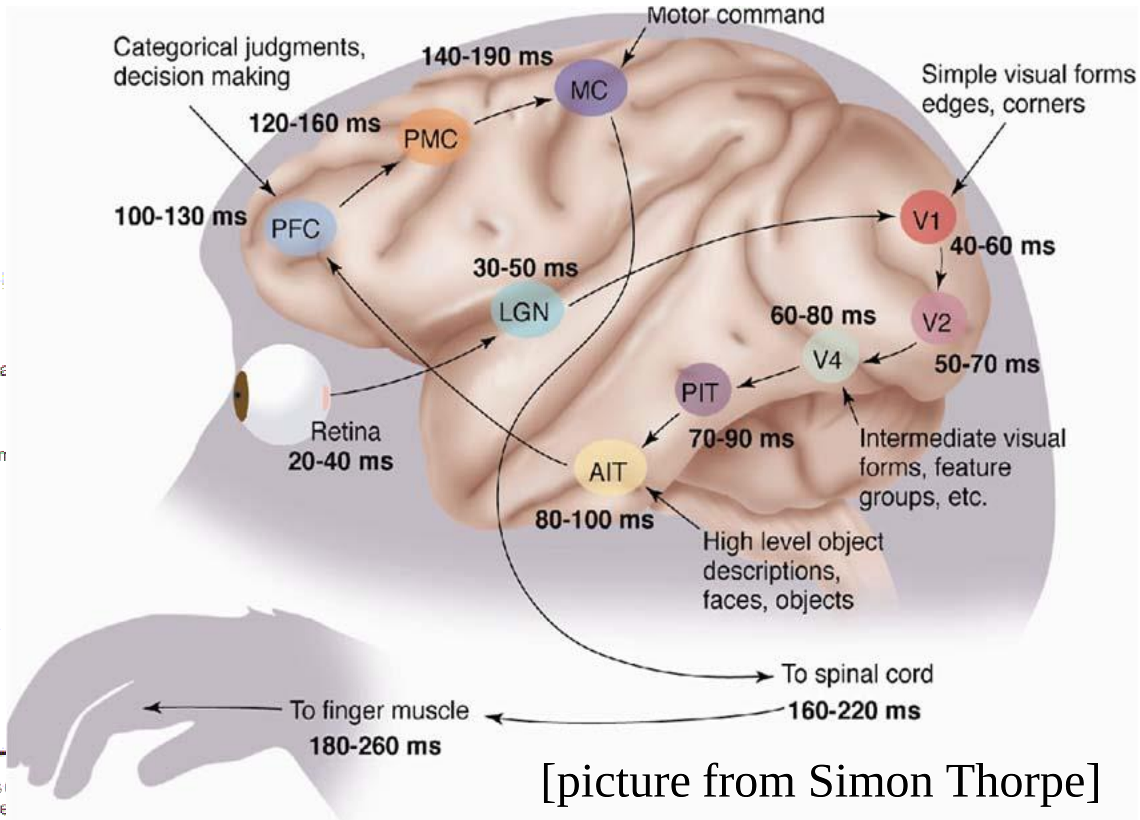


The Mammalian Visual Cortex is Hierarchical

Y LeCun
MA Ranzato

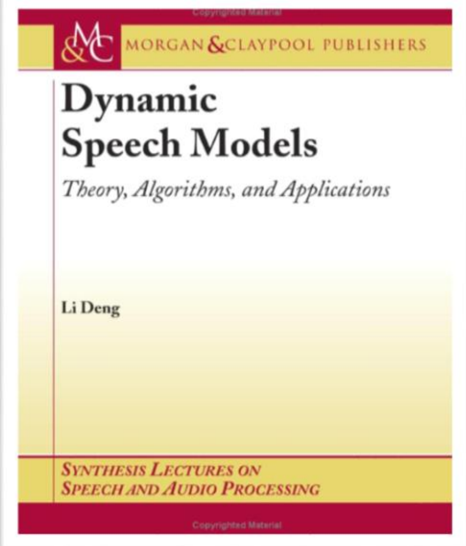
- The ventral (recognition) pathway in the visual cortex has multiple stages
- Retina - LGN - V1 - V2 - V4 - PIT - AIT ....
- Lots of intermediate representations



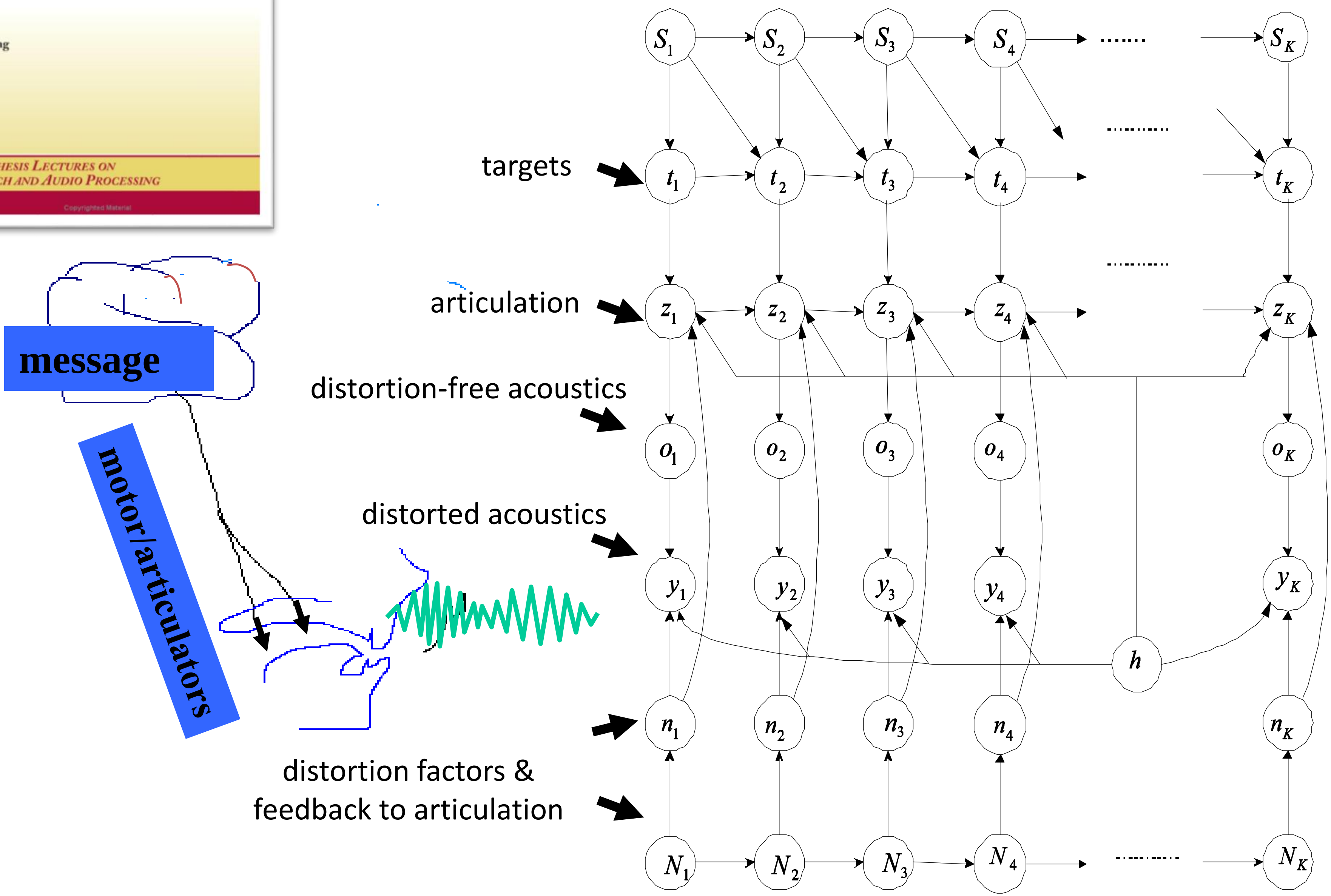


[picture from Simon Thorpe]

[Gallant & Van Essen]



(Deep) Dynamic Bayesian Net



Structured Speech Modeling

Li Deng, *Fellow, IEEE*, Dong Yu, *Member, IEEE*, and Alex Acero, *Fellow, IEEE*

Abstract—**Modeling dynamic structure of speech is a novel paradigm in speech recognition research within the generative modeling framework, and it offers a potential to overcome**

make it indistinguishable with human–human verbal interaction, at present, when users interact with any existing speech recog-

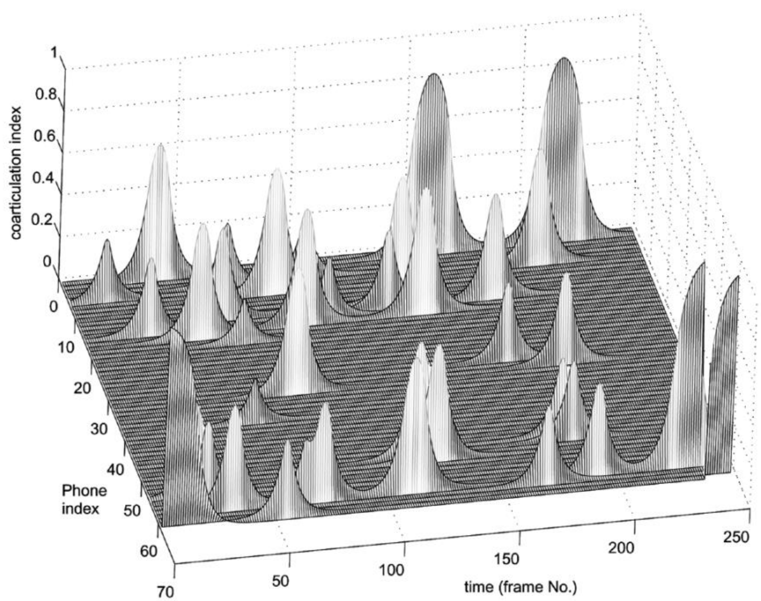
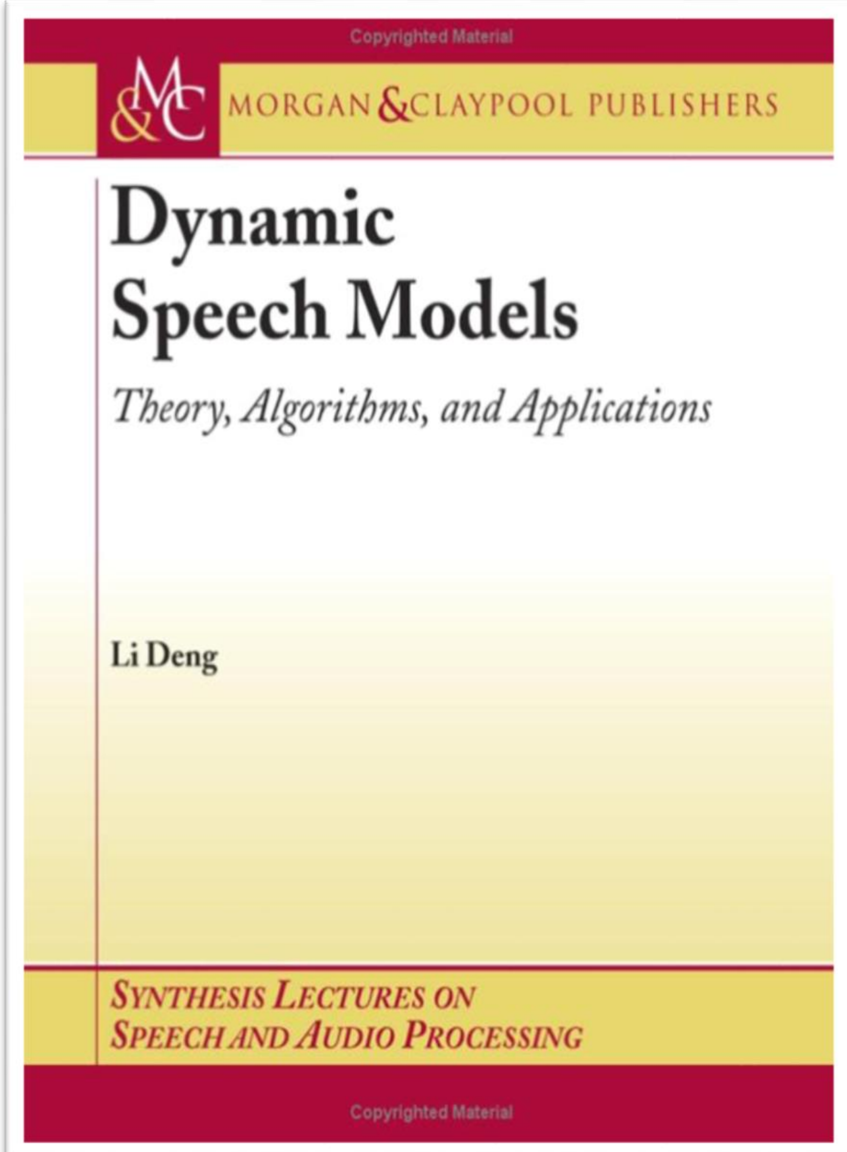


Fig. 1. Illustration of time-varying coarticulatory vectors $\mathbf{a}_k$'s for a TIMIT utterance. See text for detailed explanations.

Method	PER
CD-HMM [26]	27.3%
Augmented conditional Random Fields [26]	26.6%
Randomly initialized recurrent Neural Nets [27]	26.1%
Bayesian Triphone GMM-HMM [28]	25.6%
Monophone HTMs [29]	24.8%
Heterogeneous Classifiers [30]	24.4%
Monophone randomly initialized DNNs (6 layers)[13]	23.4%
Monophone DBN-DNNs (6 layers) [13]	22.4%
Monophone DBN-DNNs with MMI training [31]	22.1%
Triphone GMM-HMMs discriminatively trained w/ BMMI [32]	21.7%
Monophone DBN-DNNs on fbank (8 layers) [13]	20.7%
Monophone mcRBM-DBN-DNNs on fbank (5 layers) [33]	20.5%
Monophone convolutional DNNs on fbank (3 layers) [34]	20.0%

(Hidden) Dynamic Models

- Many types of dynamic models since 90's
- Good survey article on earlier work (Ostendorf et al. 1996)
- Hidden Dynamic Models (HDM/HTM) since late 90's
- This is "deep" generative model with >2 layers
- More recent work: book 2006
- Pros and cons of different models
- All intended to create more realistic speech models "deeper" than HMM for speech recognition
- But with different assumptions on speech dynamics
- How to embed such dynamic properties into the DNN framework?



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DBN (Deep) vs. DBN* (Dynamic)

- DBN-DNN (2009-2012) vs. HDM/HTM (1990's-2006)
- Distributed vs. local representations
- Massive vs. parsimonious parameters
- Product of experts vs. mixture of experts
- Generative-discriminative hybrid vs. generative models
- Longer windows vs. shorter windows

- **A neat way of "pre-training" RNN by HDM and then "fine-tuning" RNN by backprop (non-trivial gradient derivation and computation)**

Building Dynamics into Deep Recurrent Models

- (Deep) recurrent neural networks for ASR: both **acoustic** and language modeling
  - generic temporal dependency
  - lack of constraints provided by hidden speech dynamics
  - Information redundancy & inconsistency: long windows for each "frame" introducing undesirable "noise"
  - Need to go beyond unconstrained temporal dependence and ESN (while easier to learn)
- An active and exciting research area to work on

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Outline

PART I: Basics of Deep Learning (DL)

(including impact and recent history of DL (Deep Neural Net, DNN) in speech recognition)

PART II: Deeper Substance of DL

---Example 2: incorporating domain knowledge:

Speech invariance/variability vs. phonetic discrimination in Conv. NN

A Deep Convolutional Neural Net Using Heterogeneous Pooling to Tradeoff Acoustic Invariance w. Phonetic Distinction

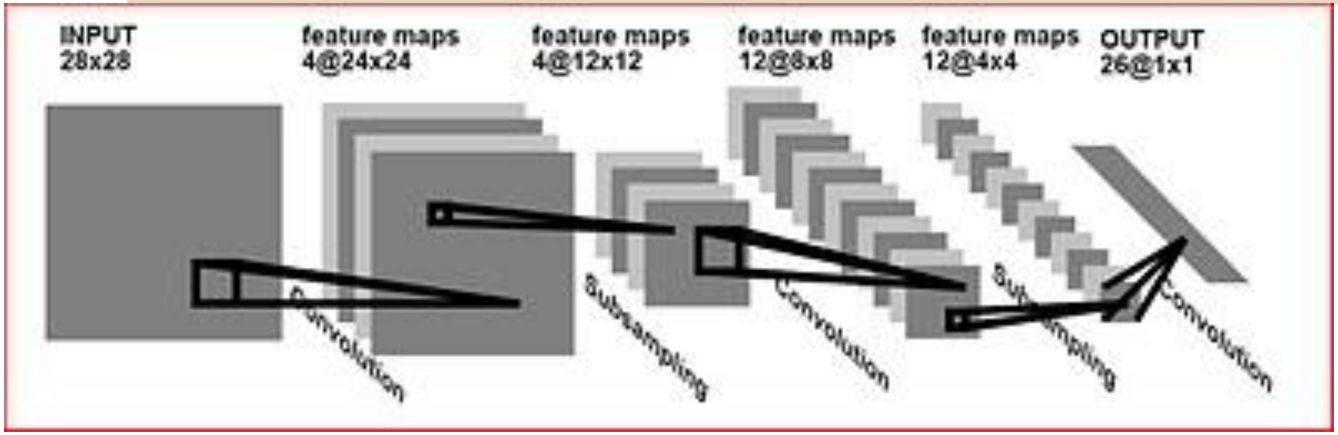
Li Deng, Ossama Abdel-Hamid, and Dong Yu

Microsoft Research, Redmond

York University, Toronto

ICASSP, May 28, 2013

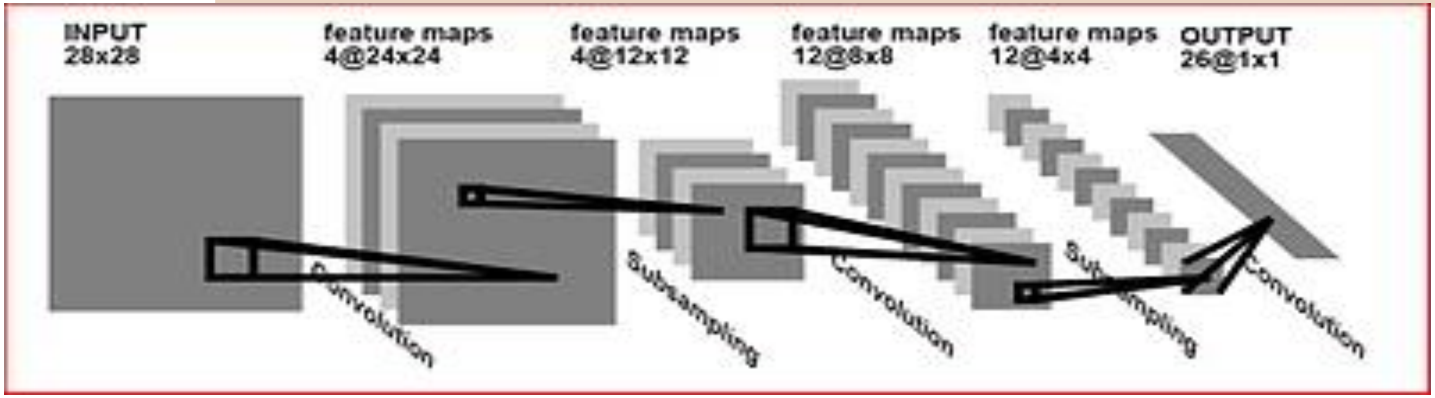
Background: Convolutional Nets (CNN)



LeCun et al. 90's

- **Convolution layer** (w. tying weights): a.k.s. "time/spatial"-invariant FIR filter
- Gives maps of replicated features; neural activities "equivariant" to translation
- **Pooling layer** (max of neighboring units in conv layer): Data reduction & some degree of **invariance.**
- 2D deep-CNN: State of the art in object recognition (Krizhevsky et al., 2012; LeCun et al.; Ciresan et al.)

Background: Convolutional Nets (CNN)



- Difficulties of CNN:
  - 2D Images: Information lost about the precise positions of parts → object confusion
  - 2D Speech spectrogram: spectral-temporal information lost about phonetic distinction
  - E.g. 1-D CNN along freq axis (Abdel-Hamid et al., 2012): (TDNN & TF-trajectory CNN)
    local weight sharing + max pooling over a range→ invariance to freq shift
    (VTL normalization)

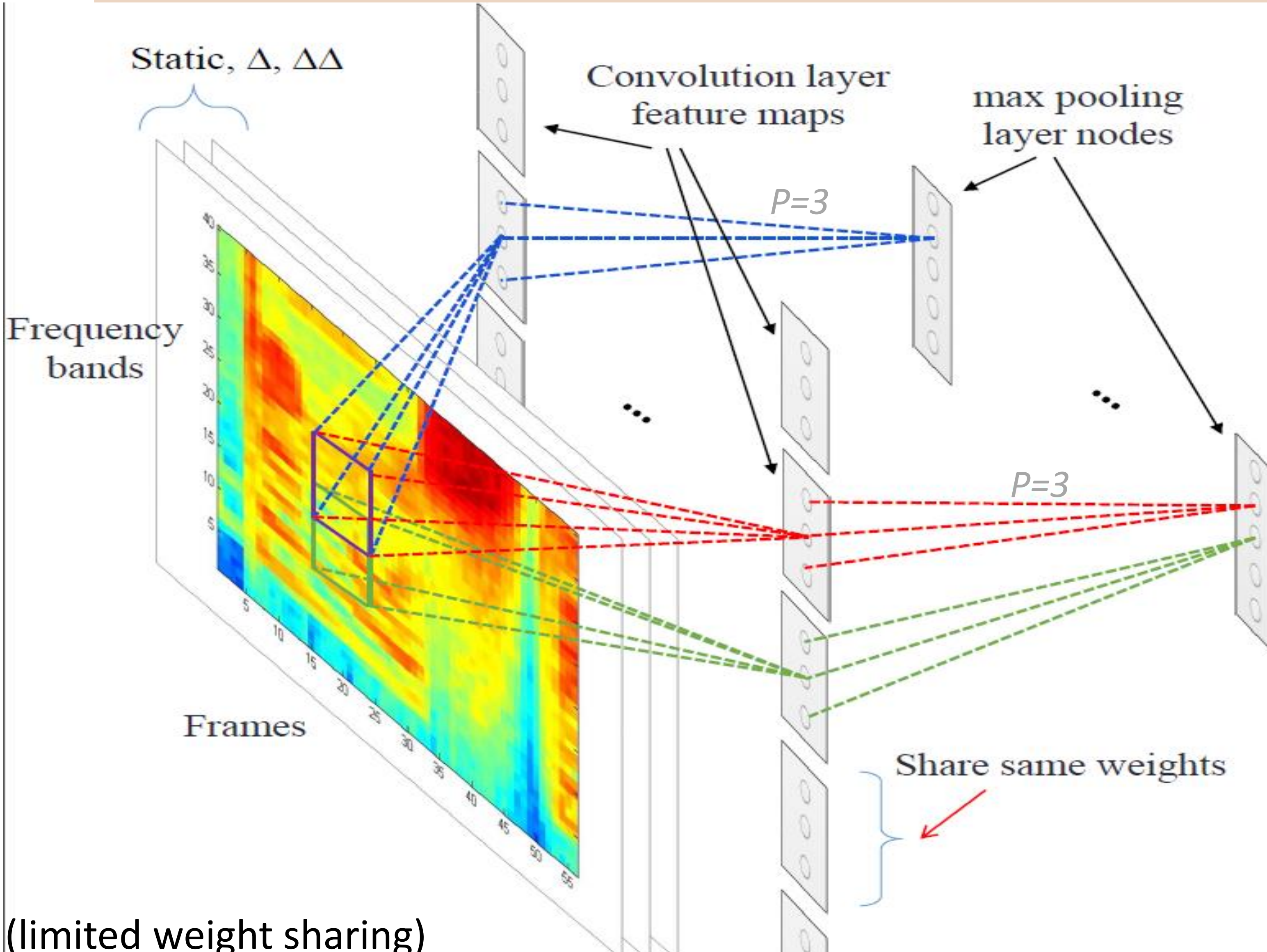
  But if freq range too small → not enough VTL normalization (acoustic invariance)
  too large → formant patterns of a sound shift → phone confusion
- Solutions for image recognition: (tried some for speech, no clear success)
  - Transforming autoencoder (Hinton et al., 2011)
  - Tiled CNN (Le et al., 2012)
  - Deconvolutional nets (Zeiler et al., 2011)
- A **good solution for speech recognition** is surprisingly simple

Main Ideas of This Paper

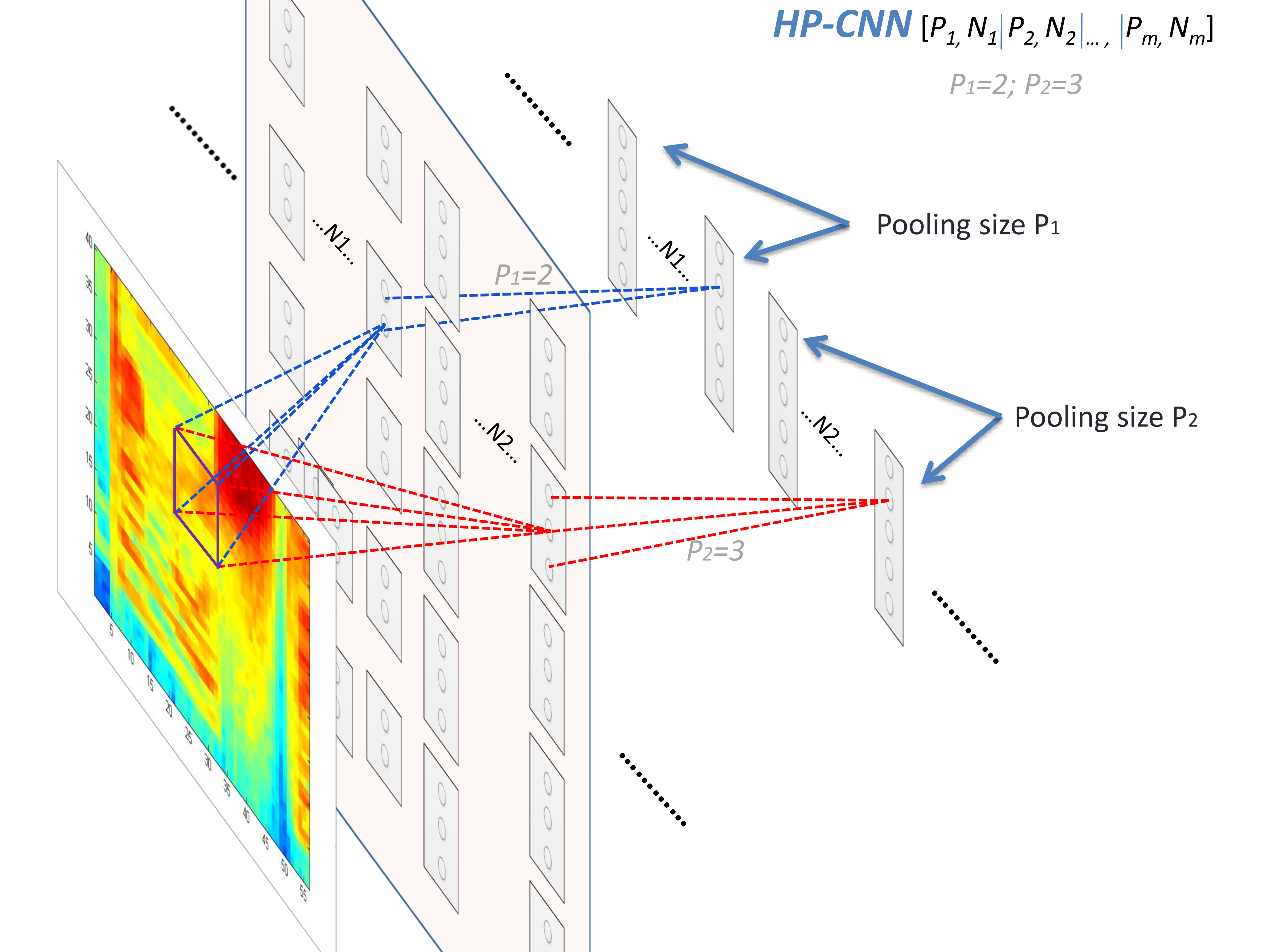
- Bring "confusion" into designing CNN intended for "invariance"
- Exploit the knowledge of how increasing the degree of invariance (to shift along frequency-axis) may reduce phonetic discrimination
- **(Kai Yu this morning: Spatial Pyramid Matching for vision)**
- Examine/predict how the pooling size (i.e. range of freq-shift invariance) affects phonetic classification errors
  - Theoretic guidance possible; e.g.
  - Phonetic reduction (in casual, conversation speech) shrinks formant space →tradeoff towards "distinction" from "invariance"→ smaller pooling size
- Use of many feature maps (afforded by CNN weight tying)
- Different pooling sizes (heterogeneous pooling) for different feature maps
  - Design and use **a distribution of pooling sizes** and randomly sample it.
  - Special case: use a fixed pooling size, optimized by validation or predicted by acoustic-phonetic "theory" (consistent for TIMIT; not as good as HP)

CNN with a Fixed Pooling Size

*(a special case of **HP-CNN** w. P=3)*



(limited weight sharing)


HP-CNN [P1, N1| P2, N2|..., |Pm, Nm]
P1=2; P2=3
Pooling size P1
Pooling size P2
...N1...
...N2...
P1=2
P2=3

Regularizing HP-CNN with “Dropout”

- A variant of the Dropout method for DNN (Hinton et al., 2012)
- Dropout in both conv and pooling layers of CNN is helpful, in addition to fully-connected DNN layers
- Dropout in the input layer (filterbanks) is not helpful
- In TIMIT, for CNN w. N=100 feature maps, and DNN hid=2000, the best dropout rate=0.2
- With dropout rate=0.5 & DNN hid=5000, error rate increases

Standard TIMIT Task: Core Testset Results

Systems	Phone Error Rate
DNN (fully-connected 5 layers)	22.3%
CNN-DNN; P=1 (2 CNN & 3 DNN layers)	**21.8%**
CNN-DNN; P=12	20.8%
CNN-DNN; P=6 (fixed P, optimal)	**20.4%**
CNN-DNN; P=6 (add dropout)	19.9%
CNN-DNN; P=1:m (HP, m=12)	**19.3%**
CNN-DNN; above (add dropout)	**18.7%**

CNN-DNN; P=1 → equivariance: 21.8% > **20.4%** (invariance w. fixed, optimal pooling size=6)
CNN-DNN; P=1:12 → Heterogeneous pooling: **19.3% <** 20.4%
Dropout is always helpful (thanks Geoff!): **18.7% <** 19.3% **;** **19.9% <** 20.4%
18.7% WAS the record low error rate on this standard task (until this morning by LSTM-RNN)

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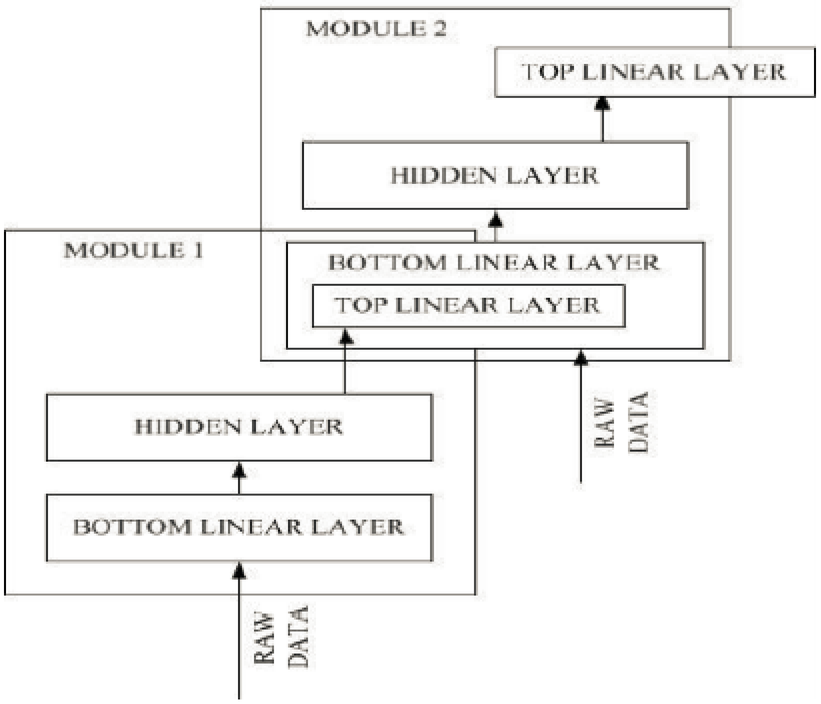
--- A few new, promising DL architectures

Google

Deep Convex Networks

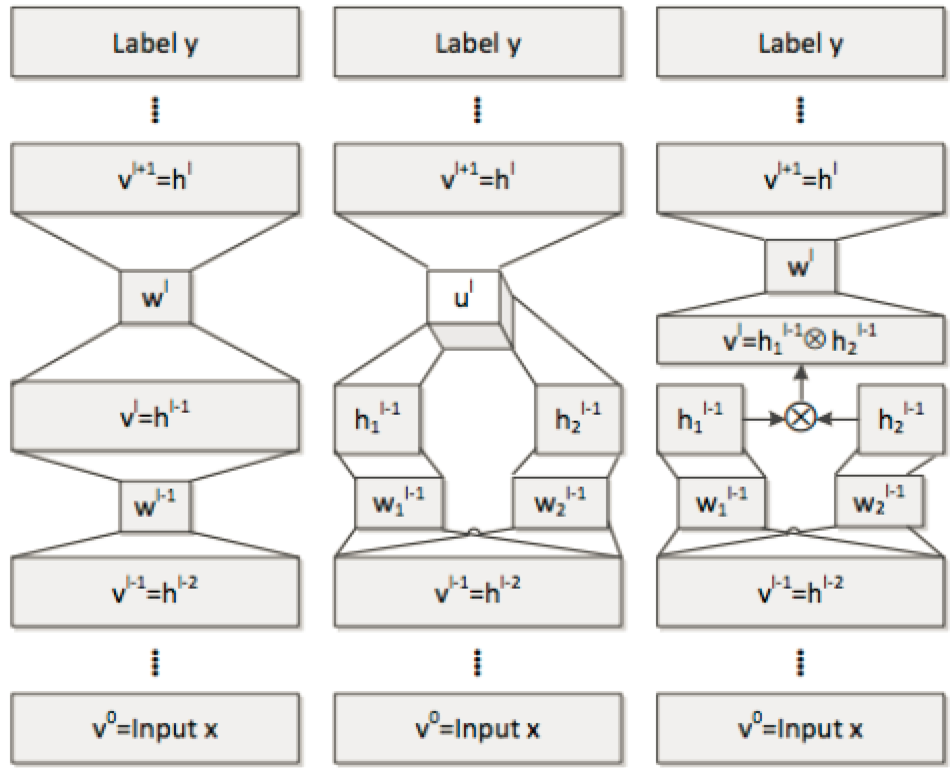
- A simple approach to build a deep model using only convex optimization techniques.
- Successfully 'convexifying' the problem is an interesting line of research.
- Very competitive and fast to train.
- So far, best performance still obtained with non-convex fine tuning and many more layers than DNNs.



Li Deng and Dong Yu. "Deep convex net: a scalable architecture for speech pattern classification." Interspeech 2011.

Deep Tensor Networks

- One example of several attempts at incorporating multiplicative nodes into deep networks.
- Very promising area of research attempting to factor out 'style' (speaker, environment) from 'content' (phonetic label) using multiplicative gating interactions.



 in print, August, 2013

Tensor Deep Stacking Networks

Brian Hutchinson, *Student Member, IEEE,* Li Deng, *Fellow, IEEE,* and Dong Yu, *Senior Member, IE*

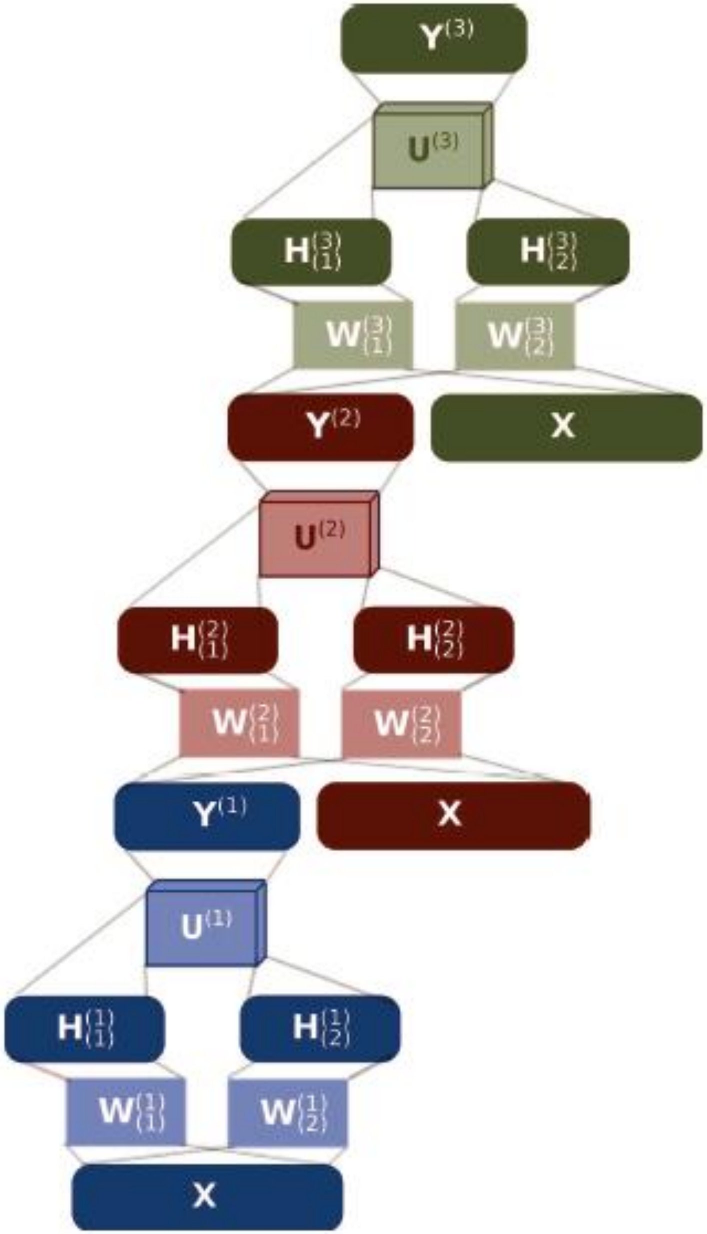
Abstract—A novel deep architecture, the Tensor Deep Stacking Network (T-DSN), is presented. The T-DSN consists of multip stacked blocks, where each block contains a bilinear mapping from two hidden layers to the output layer, using a weight tensor incorporate higher-order statistics of the hidden binary ($[0, 1]$) features. A learning algorithm for the T-DSN's weight matrices a tensors is developed and described, in which the main parameter estimation burden is shifted to a convex sub-problem with a close form solution. Using an efficient and scalable parallel implementation for CPU clusters, we train sets of T-DSNs in three popu tasks in an increasing order of the data size: handwritten digit recognition using MNIST (60k), isolated state/phone classification a continuous phone recognition using TIMIT (1.1m), and isolated phone classification using WSJ0 (5.2m). Experimental results in three tasks demonstrate the effectiveness of the T-DSN and the associated learning methods in a consistent manner. In particula a sufficient depth of the T-DSN, a symmetry in the two hidden layers structure in each T-DSN block, our model parameter learni algorithm, and a softmax layer on top of T-DSN are shown to have all contributed to the low error rates observed in the experiments all three tasks.

Index Terms—Deep learning, stacking networks, tensor, bilinear models, handwriting image classification, phone classification a recognition, MNIST, TIMIT, WSJ

INTRODUCTION

ECENTLY, a deep classification architecture built

the T-DSN retains the same linear-nonlinear interl structure as DSN in building up the deep architec



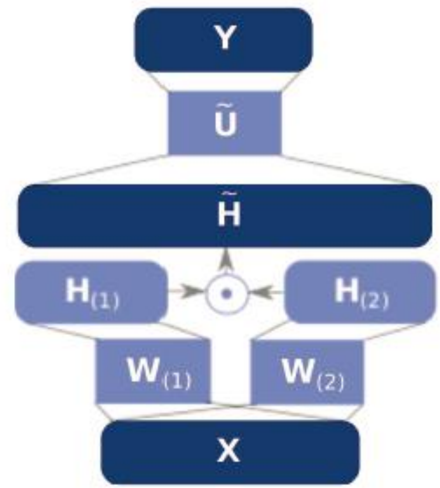


Fig. 2. Equivalent architecture to the bottom block of Fig. 1, where the tensor is unfolded into a large matrix.

Fig. 1. An example T-DSN architecture with three stacking blocks, where each block consists of three layers, and

Deep Stacking Networks for Information Retrieval

Li Deng, Xiaodong He, and Jianfeng Gao

Microsoft Research, Redmond

ICASSP, May 30, 2013

Outline

- Motivation: deep learning for Information Retrieval (IR)
  - Learning to rank
  - Semantic feature extraction for ranking
- Deep Stacking Net (DSN)
  - Basic modular architectures
  - Novel discriminative learning algorithm
- Applying DSN for IR --- learning to rank
  - Formulating IR as a classification problem
  - Special role of regularization
- Experiments
  - IR task, data sets, and features
  - Relationship between NDCG score & classification error rate
  - NDCG results on an IR task (Ads selection)

Background of IR

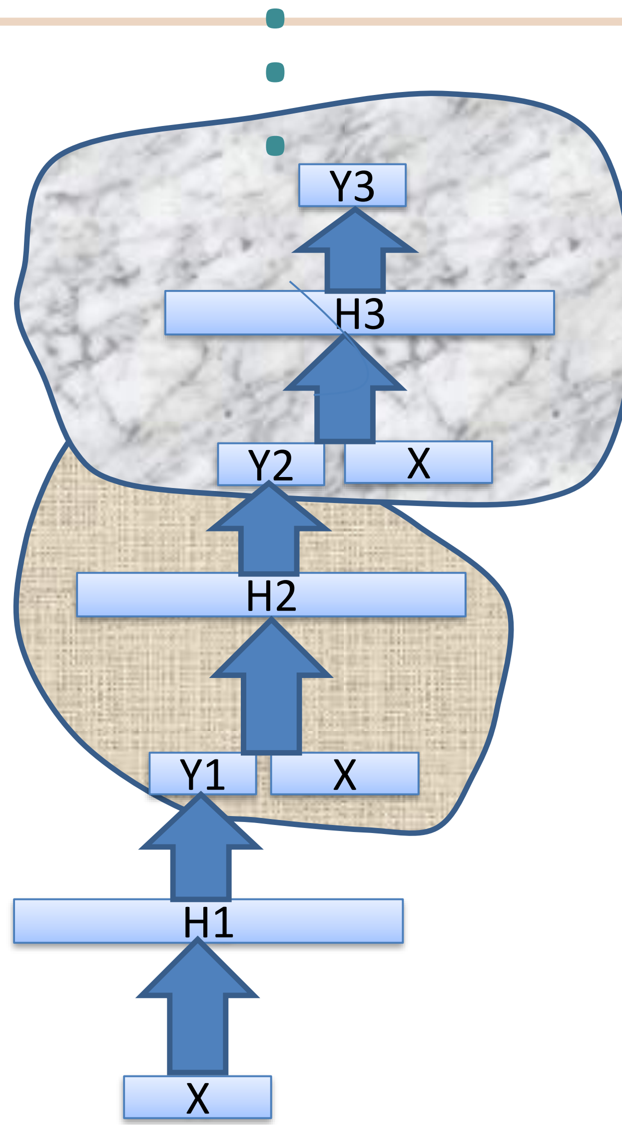
- Goal of IR: ranking text documents (D) for a query (Q)
- Common methods:
  - Lexical matching: suffers from text discrepancy btwn Q and D (e.g. vocabulary, word usage, expression style, etc.)
    - E.g., TF-IDF weighted vector space model
  - Semantic matching: to bridge lexical gaps btw Q and D
    - E.g., Latent Semantic Analysis (LSA), PLSA, LDA, etc.
  - Learning Q-D matching using clickthrough data
    - E.g., translation models, bilingual topic models etc.
  - These linear models suffer from restricted expressive power

Deep Learning for IR

- Multilayers of nonlinearities
  – Greater expressive power
  – Better able to capture semantic contents in Q and D
    - E.g., semantic hashing (Hinton et al, 2007)
  – More effective use of supervised clickthrough data
- Use of (labeled) clickthrough data for IR ranking
  – Shallow linear models: Gao et al., 2010;2011
  – Shallow nonlinear models: Burges et al., 2005;2006

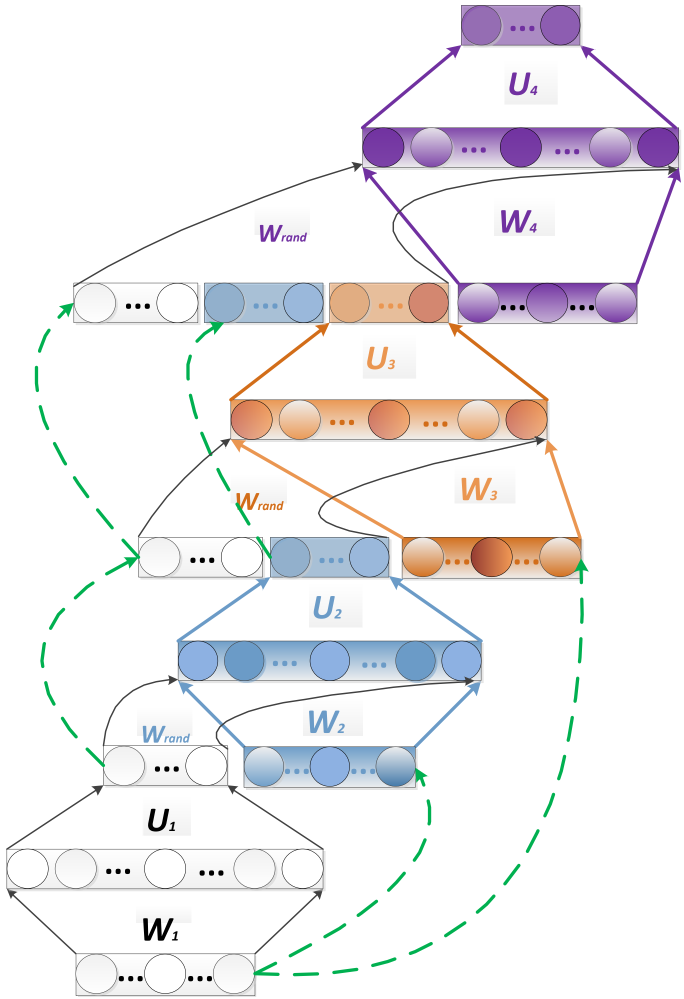
Deep Stacking Net (DSN)

- Deep Stacking Nets (Deng & Yu, Interspeech'10; Deng, Yu, Platt, ICASSP'12)
- Interleave linear/nonlinear layers
- Exploit closed-form constraints among network's weights
- Much easier to learn than DNN
- Naturally amenable to parallel training
- (Largely) convex optimization
- Extended to tensor version (Hutchinson et al, ICASSP'12, TPAMI-2013)
- Extended to kernel version (Deng et al, SLT'12)
- Works very well for MNIST, TIMIT, WSJ, SLU
- This paper: a more recent application to **IR ranking**



Learning DSN Weights --- Main Ideas

- Learn weight matrices U and W in individual modules separately.
- Given W and linear output layer, U can be expressed as explicit nonlinear function of W.
- This nonlinear function is used as the constraint in solving nonlinear least square for learning W.
- Initializing W with RBM (bottom layer)
- For higher layers, part of W is initialized with the optimized W from the immediately lower layer and part of it with random numbers

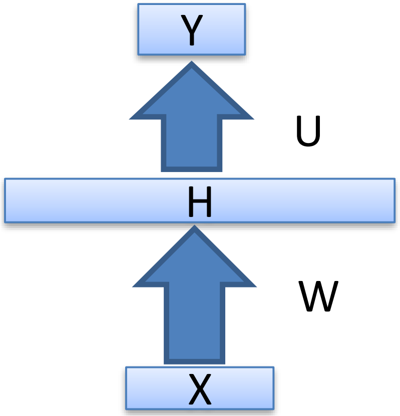


Learning DSN Weights --- Single Module

$$\mathrm{E} = \frac{1}{2}\sum_{n} ||\boldsymbol{y}_n - \boldsymbol{t}_n||^2, \qquad \text{where } \boldsymbol{y}_n = \boldsymbol{U}^T\boldsymbol{h}_n = \boldsymbol{U}^T\sigma(\boldsymbol{W}^T\boldsymbol{x}_n) = G_n(\boldsymbol{U},\boldsymbol{W})$$

$$\frac{\partial E}{\partial \boldsymbol{U}} = 2\boldsymbol{H}(\boldsymbol{U}^T\boldsymbol{H} - \boldsymbol{T})^T \rightarrow \quad \boldsymbol{U} = \left(\boldsymbol{H}\boldsymbol{H}^T\right)^{-1}\boldsymbol{H}\boldsymbol{T}^T = \mathrm{F}(\boldsymbol{W}), \text{ where } \boldsymbol{h}_n = \sigma(\boldsymbol{W}^T\boldsymbol{x}_n)$$

$\mathrm{E} = \frac{1}{2}\sum_n ||G_n(\boldsymbol{U},\boldsymbol{W}) - \boldsymbol{t}_n||^2$, *subject to* ***U***$= \mathrm{F}(\boldsymbol{W})$,



Use of Lagrange multiplier method:

$$\mathrm{E} = \frac{1}{2}\sum_n ||G_n(\boldsymbol{U},\boldsymbol{W}) - \boldsymbol{t}_n||^2 + \lambda\, ||\boldsymbol{U} - \mathrm{F}(\boldsymbol{W})||$$

to learn ***W*** and then ***U*** → no longer backpropagation

- Advantages found:
  --- less noise in gradient than using chain rule ignoring explicit constraint ***U***$= \mathrm{F}(\boldsymbol{W})$
  --- batch learning is effective, aiding parallel training

Experimental Evaluation

- IR task
  - Sponsored Search: retrieve and rank relevant ads given a query
- Data sets
  - Training: 189K query–ads pairs
  - Testing: 58K query–ads pairs
- Features to DSN
  - A total of 160 features in two categories
    - Text features: TF-IDF, word overlap, length, etc.
    - User click features: clickthrough, clicked queries, etc.
- State-of-the-art baseline system (Burges et al. 2006)
  - LambdaRank, a single-hidden-layer neural network
  - Trained to maximize (a smoothed approximation of) NDCG via heuristic lambda-function

Evaluation Metric

- Metric: Normalized Discounted Cumulative Gain (NDCG)
- DCG at rank p = $relevance_1 + \sum_{i=2}^{p} \frac{relevance_i}{\log_2 i}$; $relevance_i$: human label of doc_i, scale 0-4
- IDCG: Ideal DCG, DCG score when assuming docs are ranked by human label
- NDCG = DCG/ IDCG
- 1 NDCG pt (0.01) in our setting is statistically significant

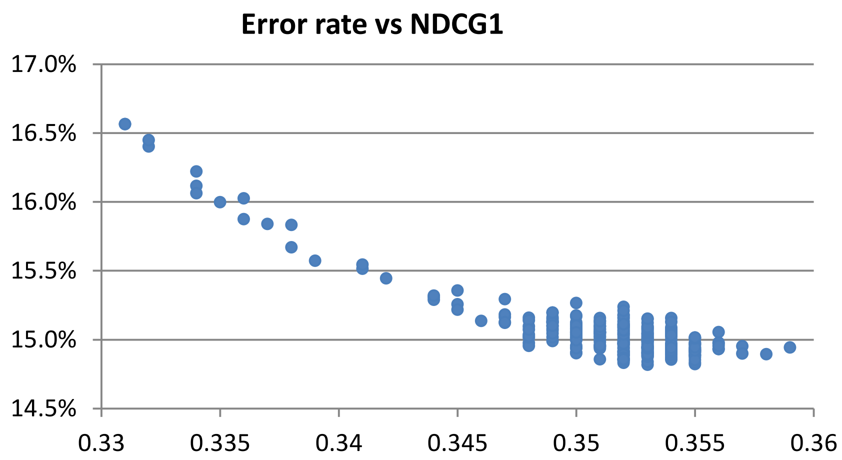
NDCG Results

IR Quality measures (NDCG) for the DSN System vs. Baseline

IR Systems	NDCG@1	NDCG@3	NDCG@10
LambdaRank	0.331	0.347	0.382
DSN system	0.359	0.366	0.402

Analysis

Relationship between classification error rates and NDCG@1 measure)



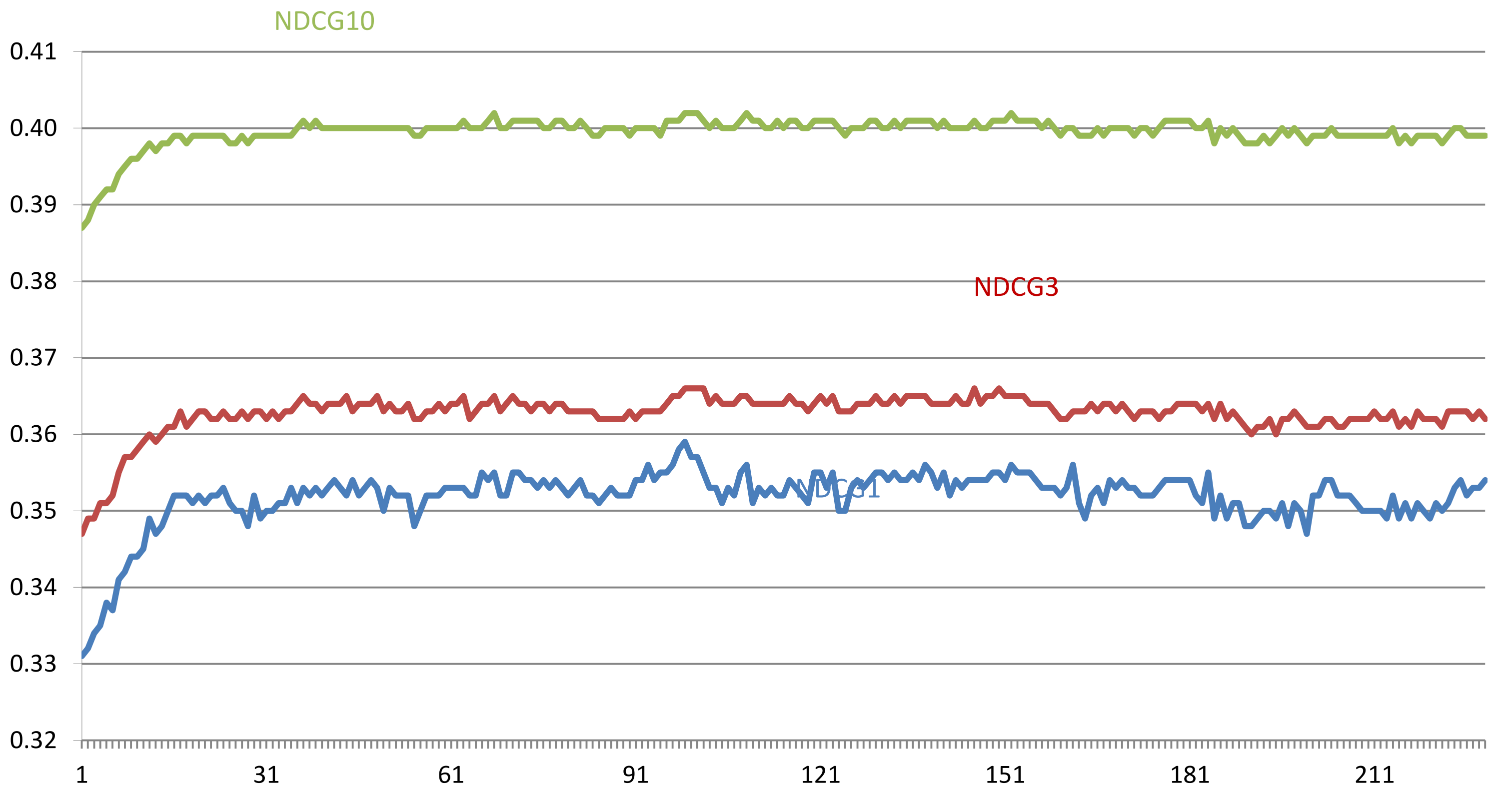
Observations:

- Correlation is clearly evidenced for NDCG1 < 0.35
- Weaker correlation in the high IR-quality region, i.e., NDCG1 > 0.35

Implication:

- Due to the inconsistency between the training objective and the IR-quality measure
- It is desirable to train the model to optimize the end-to-end IR quality directly

Learning Curves



Conclusions (of this ICASSP-2013 paper)

- First study on the use of deep learning techniques for learning-to-rank in IR problems
- Significantly better than shallow neural network
- Model trained by MSE
  - Generally correlated well with the NDCG as the IR quality measure
  - But weaker correlation in the region of high IR quality
- Deep learning using end-to-end IR-relevant metric is a key future direction

Microsoft Research

(including impact and recent history of DL (Deep Neural Net, DNN) in speech recognition)

Outline

PART II: Deeper Substance of DL

---Technical introduction: RBM, DBN, DNN, DNN-HMM, CNN, RNN

---Examples of incorporating domain knowledge (about speech) into DL architectures

1. Hidden/articulatory Speech dynamics into RNN
2. Speech invariance/class-discrim.into deep-CNN

-A few new, promising DL architectures **(CONTINUED)**

New Types of Deep Neural Network & Learning for Speech Recognition+
An Overview

Li Deng, Geoffrey Hinton, Brian Kingsbury

MSR, U. Toronto/Google, IBM

ICASSP Special Session, May 28, 2013

Microsoft UNIVERSITY OF TORONTO Google IBM Research

Special Session Motivations

- Huge impact of deep neural nets (DNN) in speech (and vision, language, etc.)

The New York Times

Special Session Motivations

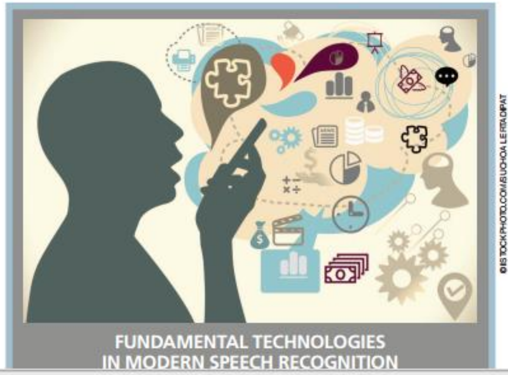
- Review article (2011-2012)
- Key factors:
  - Deeper network
  - Faster hardware
  - Larger network output layer (& hidden, input layers)
  - Better network initialization (not essential with big data)
- Rather standard MLP architecture
- Also standard backprop learning (1980’s)

IEEE Sig. Proc. Mag, Nov 2012

[Geoffrey Hinton, Li Deng, Dong Yu, George E. Dahl, Abdel-rahman Mohamed, Navdeep Jaitly, Andrew Senior, Vincent Vanhoucke, Patrick Nguyen, Tara N. Sainath, and Brian Kingsbury]

Deep Neural Networks for Acoustic Modeling in Speech Recognition

[The shared views of four research groups]

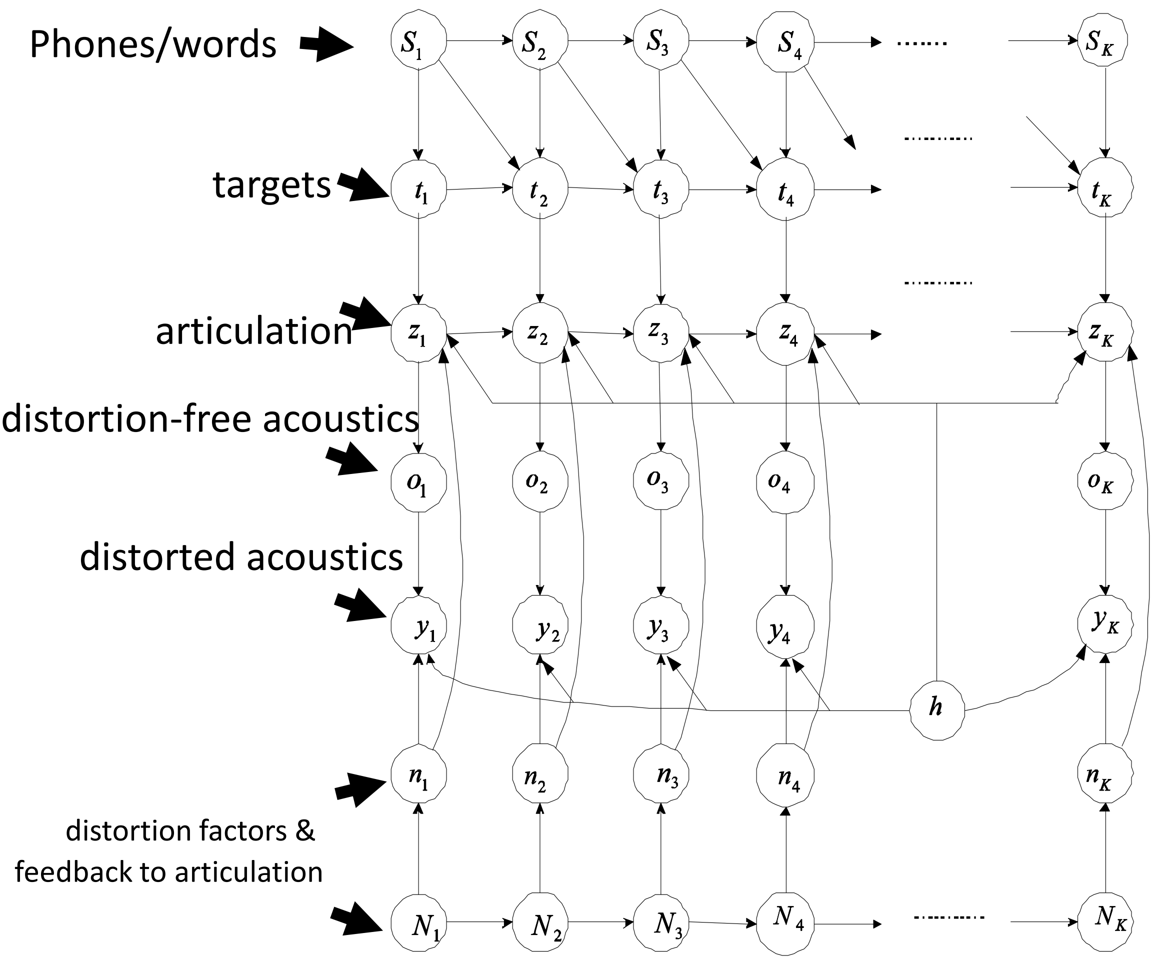


Take-Away from This Special Session

- New models and new learning methods
- Key capabilities of DNNs in knowledge transfer, learning representations, etc.
- Advances in DNNs since the SPM overview paper

Recent History of “Deep” Models in Speech

- MSR’s (deep) Dyn. Bayes Net (2004-2007)



- U Toronto’s DBN-DNN (2006-**2009**)

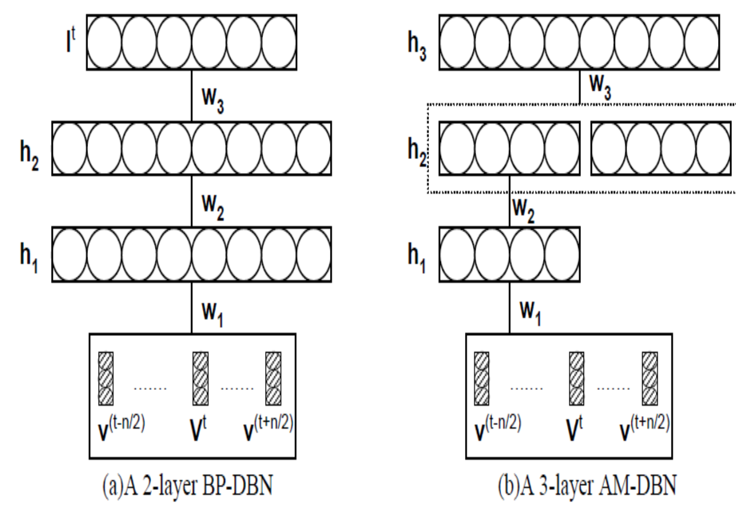
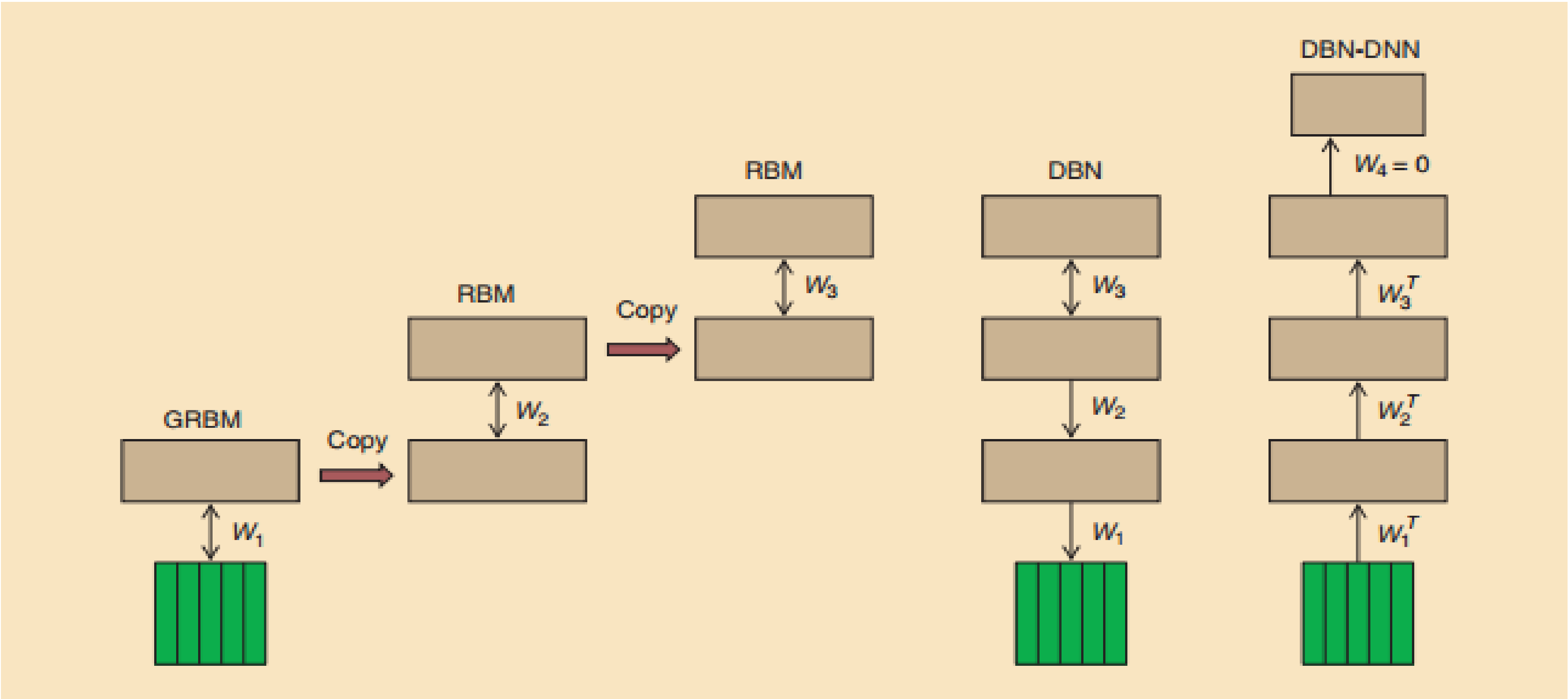


Figure 2: The DBN architectures used in this work.

Mohamed, Dahl, Hinton, NIPS-WS, 2009
(a simple “recipe”)

Hinton's 2009"Recipe"



[FIG1] The sequence of operations used to create a DBN with three hidden layers and to convert it to a pretrained DBN-DNN. First, a GRBM is trained to model a window of frames of real-valued acoustic coefficients. Then the states of the binary hidden units of the GRBM are used as data for training an RBM. This is repeated to create as many hidden layers as desired. Then the stack of RBMs is converted to a single generative model, a DBN, by replacing the undirected connections of the lower level RBMs by top-down, directed connections. Finally, a pretrained DBN-DNN is created by adding a "softmax" output layer that contains one unit for each possible state of each HMM. The DBN-DNN is then discriminatively trained to predict the HMM state corresponding to the central frame of the input window in a forced alignment.

IEEE SIGNAL PROCESSING MAGAZINE [6] NOVEMBER 2012

Deep Learning for Phone Recognition
(a stunning discovery at MSR, 2009)

	METHOD	Error rate
	(Shallow) GMM-HMM (1987-2010)	$\boxed{27.3\%}$
	AUGMENTED CONDITIONAL RANDOM FIELDS [26]	26.6%
	RANDOMLY INITIALIZED RECURRENT NEURAL NETS [27]	26.1%
	BAYESIAN TRIPHONE GMM-HMM [28]	25.6%
	Deep/hidden trajectory model (MSR, 2006)	$\boxed{24.8\%}$
DNN	MONOPHONE RANDOMLY INITIALIZED DNNs (SIX LAYERS) [13]	23.4%
DNN	MONOPHONE DBN-DNNs (SIX LAYERS) [13]	22.4%
	MONOPHONE DBN-DNNs WITH MMI TRAINING (MSR, 2010)	22.1%
	TRIPHONE GMM-HMMs DT W/ BMMI (IBM, 2010)	21.7%
	MONOPHONE DBN-DNNs ON FBANK (EIGHT LAYERS) [13]	20.7%
	MONOPHONE MCRBM-DBN-DNNs ON FBANK (FIVE LAYERS) [33]	20.5%
	Deep convolutional nets w. DropOut & Heter. Pooling (MSR, 2012)	**18.7%**

Error pattern A

Error pattern B

IEEE SIGNAL PROCESSING MAGAZINE [6] NOVEMBER 2012

Deep Learning for Large-Vocabulary Speech Recognition

TASK	HOURS OF TRAINING DATA	DNN-HMM	GMM-HMM WITH SAME DATA	GMM-HMM WITH MORE DATA
SWITCHBOARD	309	16.1	23.6	17.1 (2,000 H)
ENGLISH BROADCAST NEWS	50	17.5	18.8	
BING VOICE SEARCH (SENTENCE ERROR RATES)	24	30.4	36.2	
GOOGLE VOICE INPUT	5,870	12.3		16.0 (>> 5,870 H)
YOUTUBE	1,400	47.6	52.3	

IEEE SIGNAL PROCESSING MAGAZINE [6] NOVEMBER 2012

New Discoveries about the DNN "Recipe" since 2009

- Pre-training not needed when a lot of labeled data are available (2010)
- The recipe works well for LVCSR when DNN output units correspond to CD HMM states (2010)
- **Decoding alg. & infrastructure largely unchanged, enabling industry-scale speech recognition (2010-2013)**
- Filterbank features (closer to waveform) better than MFCCs for DNNs (opposite to GMM systems) (2011-2013)
- DNN works surprisingly well for noisy speech (2012)
- Fully-connected DNN can be modified to include "convolutional" layers to handle speech variability (2012-2013)
- DNN highly effective for multi-task/transfer learning (e.g. multilingual ASR, 2012-2013)
- DNN effective for applications beyond ASR.

Five Technical Papers in Our Special Session

RECENT ADVANCES IN DEEP LEARNING FOR SPEECH RESEARCH AT MICROSOFT

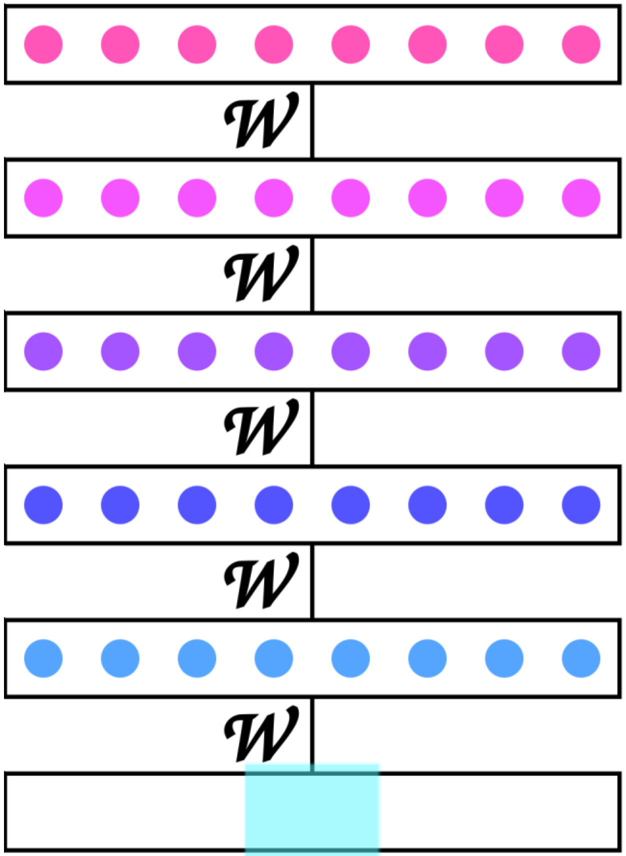
IMPROVING DEEP NEURAL NETWORKS FOR LVCSR USING RECTIFIED LINEAR UNITS AND DROPOUT

DEEP CONVOLUTIONAL NEURAL NETWORKS FOR LVCSR

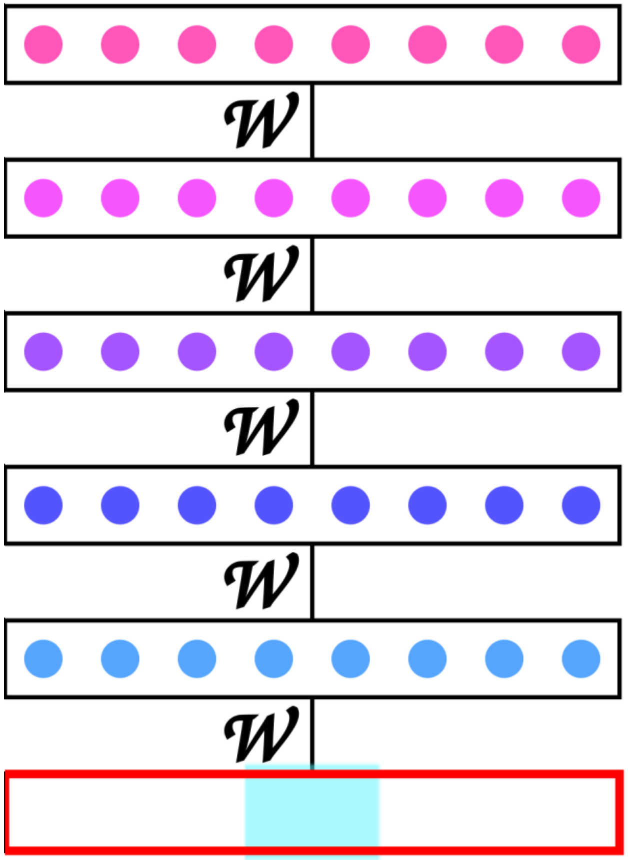
MULTILINGUAL ACOUSTIC MODELS USING DISTRIBUTED DEEP NEURAL NETWORKS

ADVANCES IN OPTIMIZING RECURRENT NETWORKS

Themes in the Session

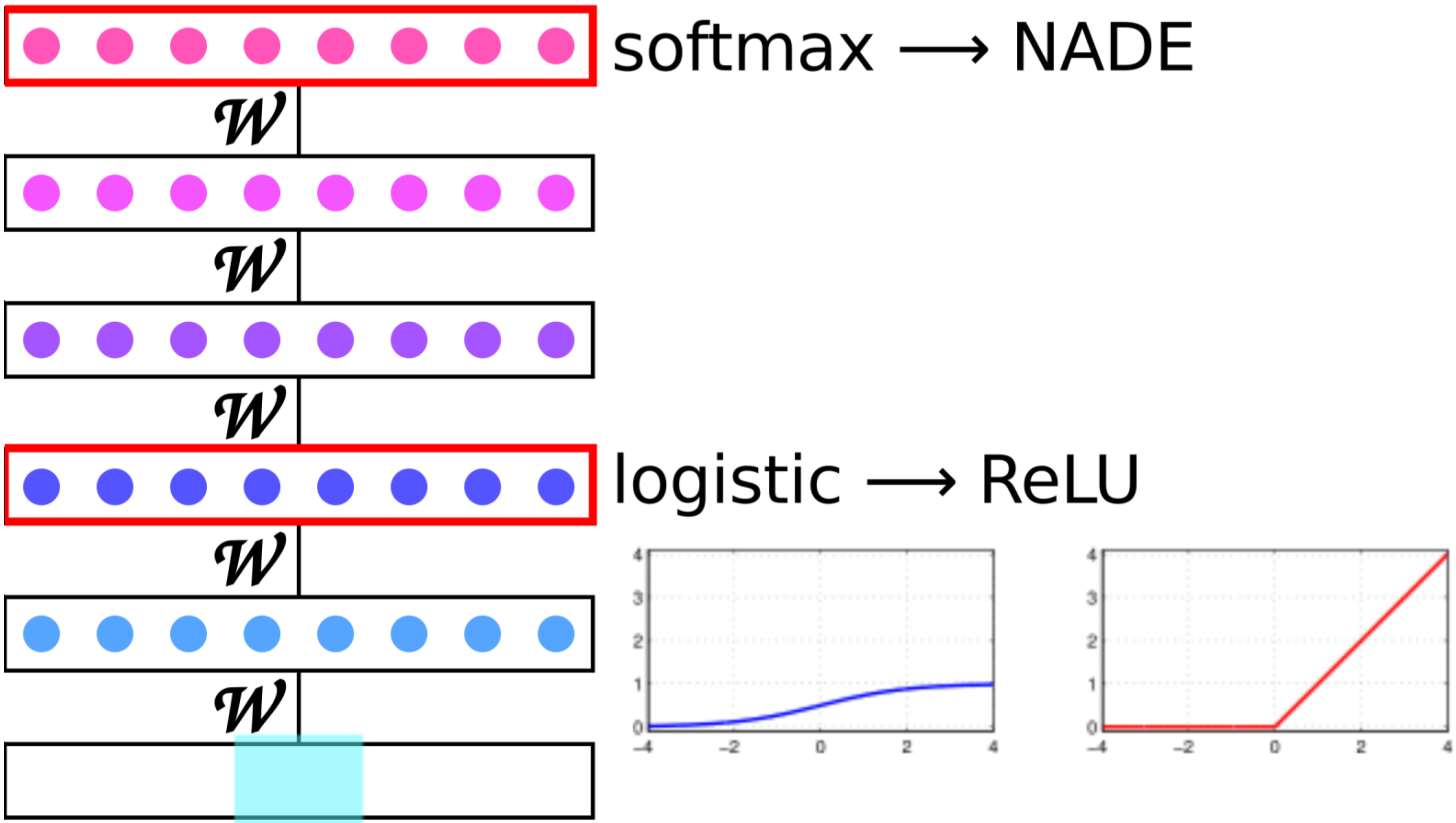


Themes: Better Inputs

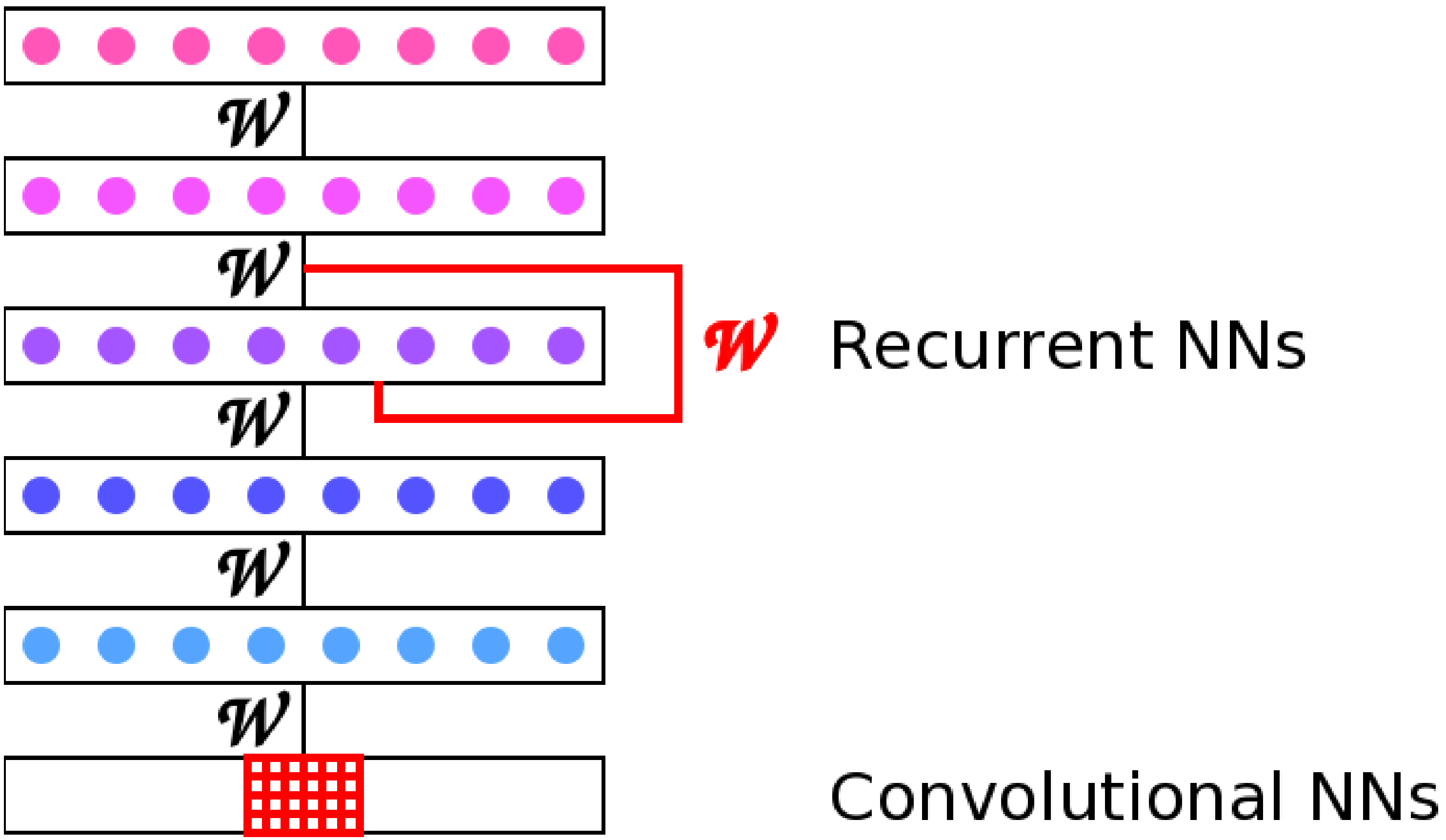


MFCCs ⟶ log Mel spectra

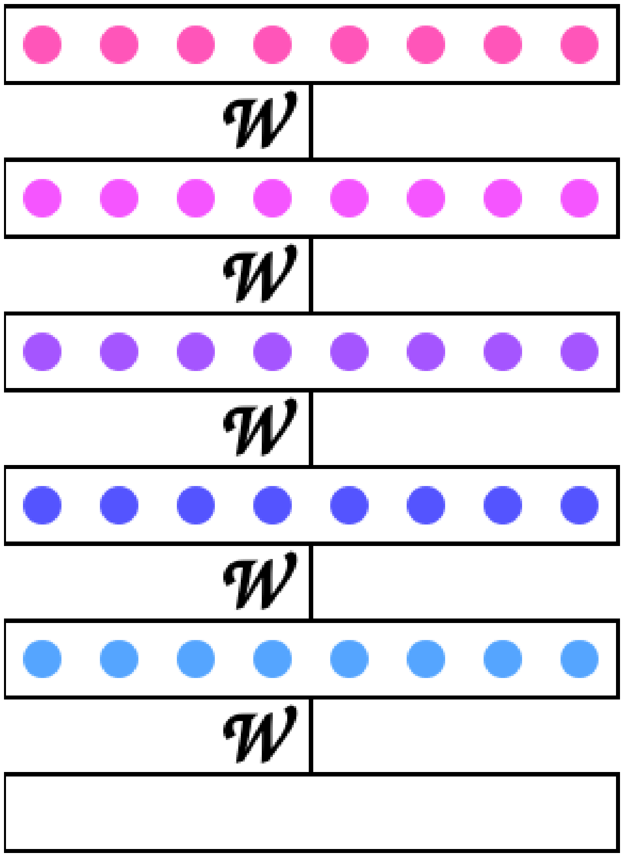
Themes: Nonlinearities



Themes: Architectures



Themes: Optimization



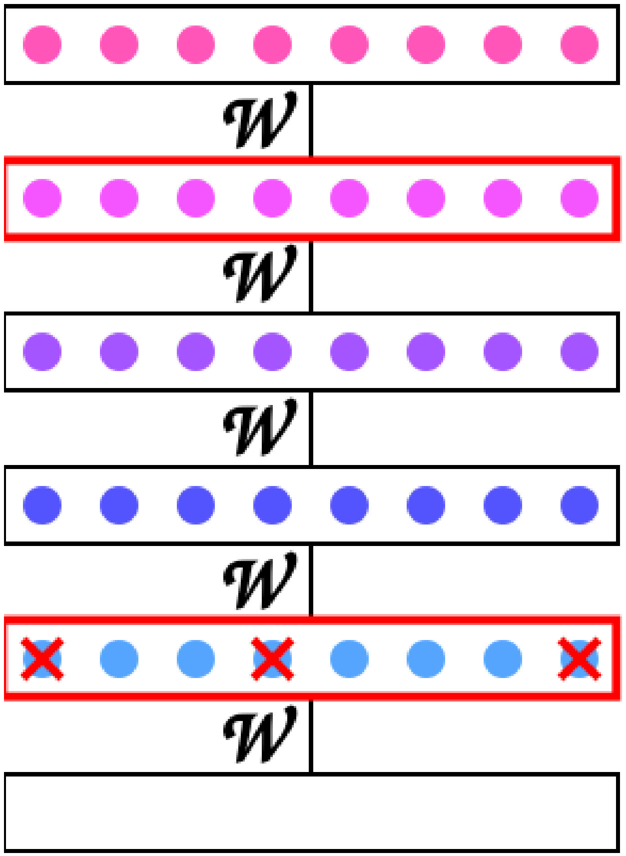
Distributed asynchronous SGD

Distributed Hessian-free optimization

Nesterov's accelerated gradient method

Gradient clipping

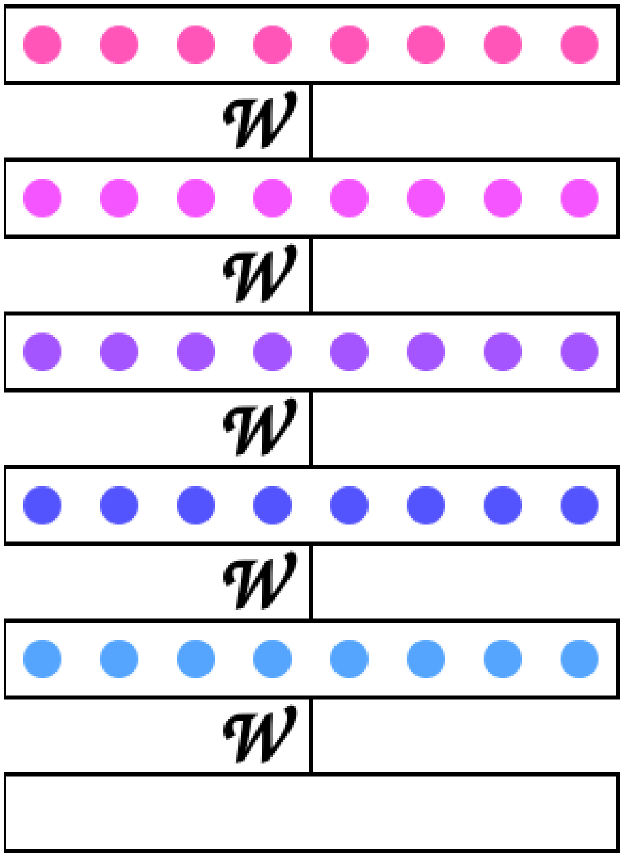
Themes: Regularization



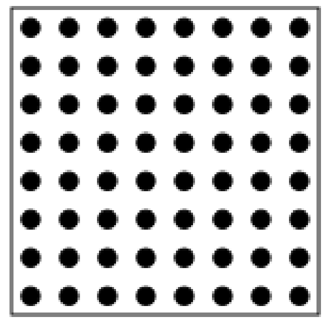
Sparsity in hidden representations

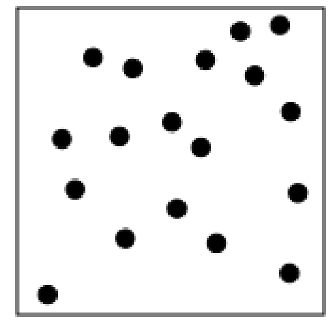
Dropout

Themes: Hyperparameters



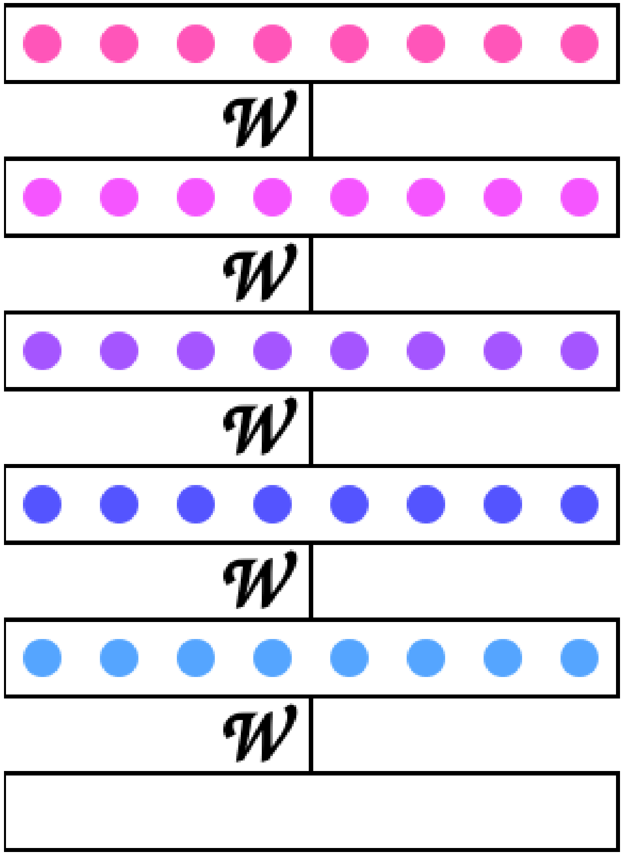
Grid search $\longrightarrow$ Sampling





Bayesian optimization

Themes: Multi-task Learning



Multi-lingual acoustic modeling

Mixed-bandwidth acoustic modeling

Recent Advances in Deep Learning for Speech Research at Microsoft

Li Deng, Jinyu Li, Jui-Ting Huang, Kaisheng Yao, Dong Yu, Frank Seide, Mike Seltzer, Geoff Zweig, Xiaodong He, Jason Williams, Yifan Gong, Alex Acero

Microsoft

Microsoft Research

ICASSP Special Session, May 28, 2013

Outline

- Advances in deep learning for **features**/ representations
- Advances in deep learning for **models/** architectures
- Systems and applications in acoustic modeling, language modeling, dialogue, (and information retrieval/search)

Learning Features/Representations

- Advances in deep learning for **features**/ representations
- Advances in deep learning for **models/** architectures
- Systems and applications in acoustic modeling, language modeling, dialogue, (and information retrieval/search)

The New York Times

Scientists See Promise in Deep-Learning Programs

John Markoff

November 23, 2012

Back to Primitive Spectral Features

- Philosophy of deep learning:
  - Learning representations automatically instead of manually engineering/design them (e.g., MFCC, PLP)
- DNN capability in representing correlated feature dimensions
- → eliminate cosine transform in MFCC in favor of filterbanks in spectral domain

Back to Primitive Spectral Features

- Philosophy of deep learning:
  - Learning representations automatically instead of manually engineering/design them (e.g., MFCC, PLP)
- DNN capability in representing correlated feature dimensions
- → eliminate cosine transform in MFCC in favor of filterbanks and spectrograms in the spectral domain

INTERSPEECH 2010

Binary Coding of Speech Spectrograms Using a Deep Auto-encoder

L. Deng[1], M. Seltzer[1], D. Yu[1], A. Acero[1], A. Mohamed[2], and G. Hinton[2]

[1] Microsoft Research, One Microsoft Way, Redmond, WA 98052, US
[2] University of Toronto, Toronto, Ontario, Canada

In early 2010, we discovered:

For deep autoencoding of speech features:

- Both spectrogram/filterbank features are better than MFCCs
- Better to use spectrogram features than filterbanks
- "Better" in terms of coding efficiency (i.e., errors/energy)

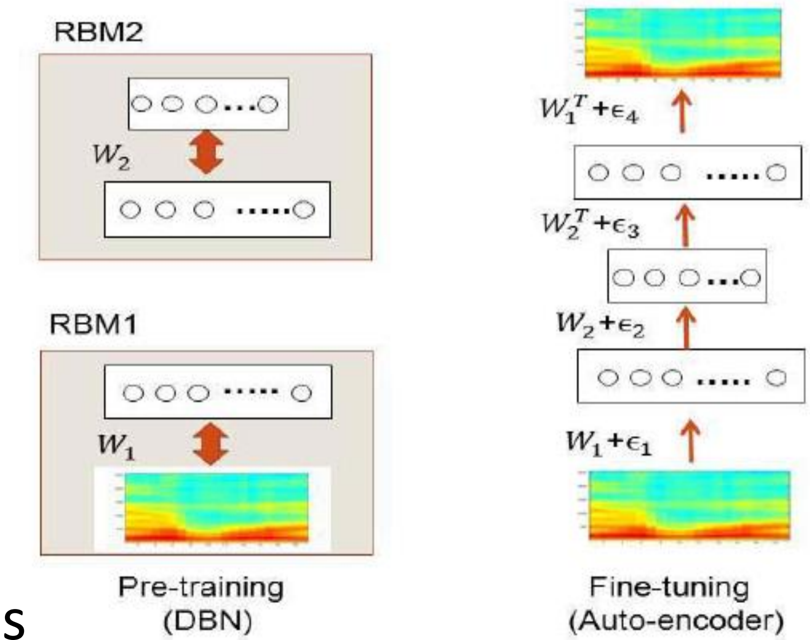


Fig.1. *Left: Illustration of pre-training of the DBN that consists of two RBMs used in this work. Right: Illustration of fine tuning that produces the final deep auto-encoder.*

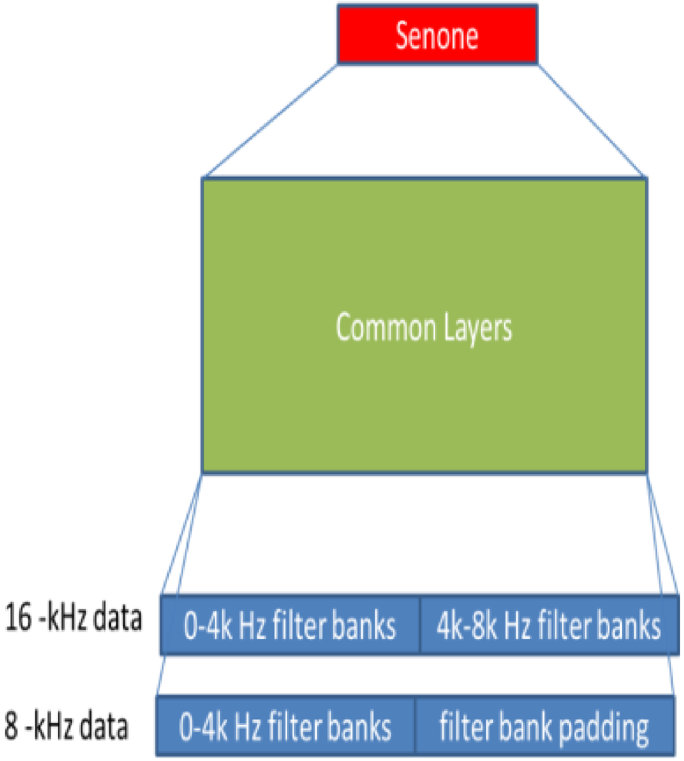
LVCSR Using Spectral Features

LVCSR Systems	Word error rate
Best GMM-HMM (MFCCs; fMPE+BMMI)	34.7%
DNN (MFCCs)	**31.6%**
DNN (Spectrogram --- 256 log FFT bins)	**32.3%**
DNN (29 log filter-banks)	30.1%
DNN (40 log filter-banks)	**29.9%**

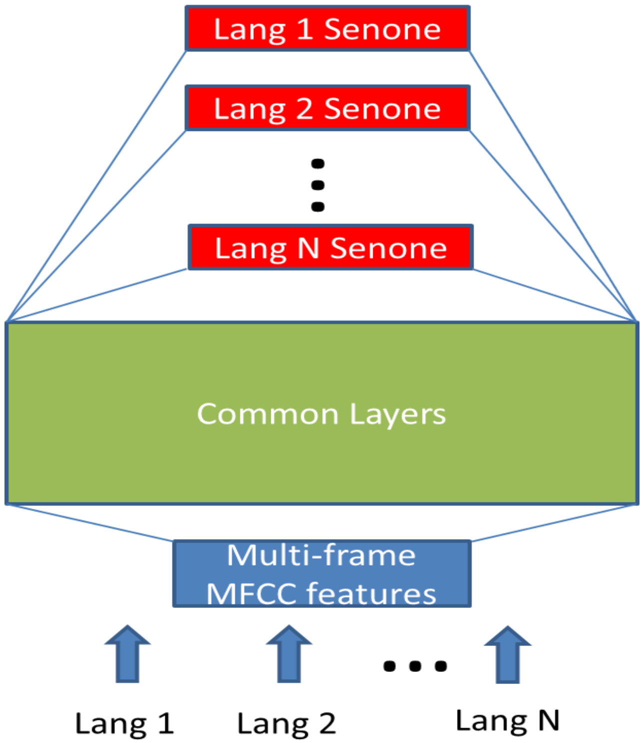
- Filterbanks > MFCC > Spectrograms
- Not quite consistent with deep autoencoder results
- Further research: regularization, online feature normalization at sentence level, etc.

Learning Multi-Task Features

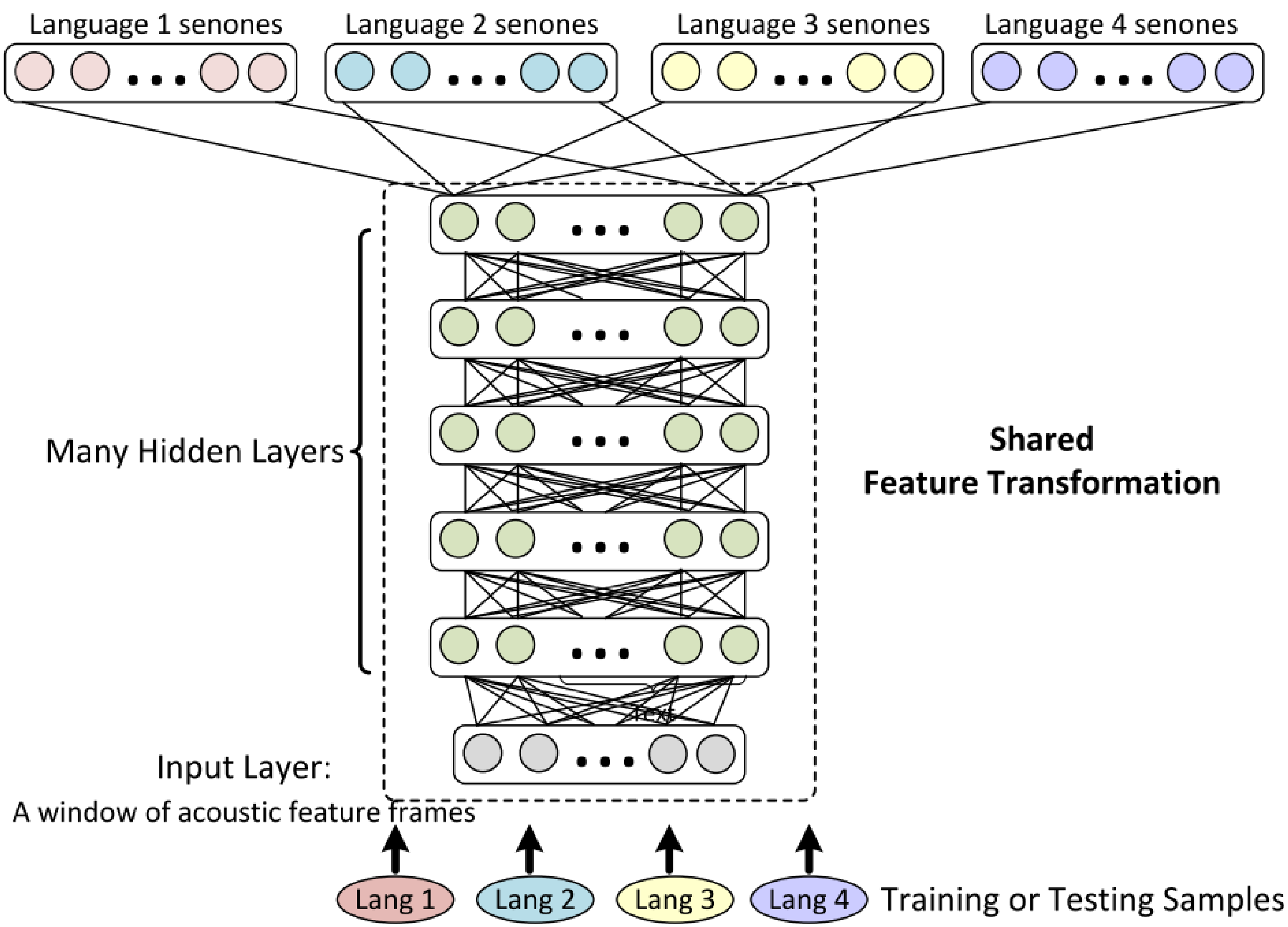
Mixed-Band DNN architecture:



Multilingual DNN architecture:



Shared Hidden Layers with Language-Specific Output Layers



Learning Multi-Task Features

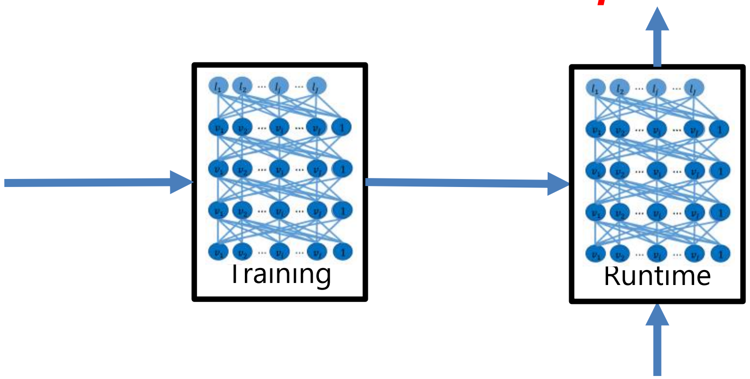
8KHz waveforms

16KHz waveforms

English training data

French training data

Chinese training data



English ***or French*** ***or Chinese*** words

with fewer errors

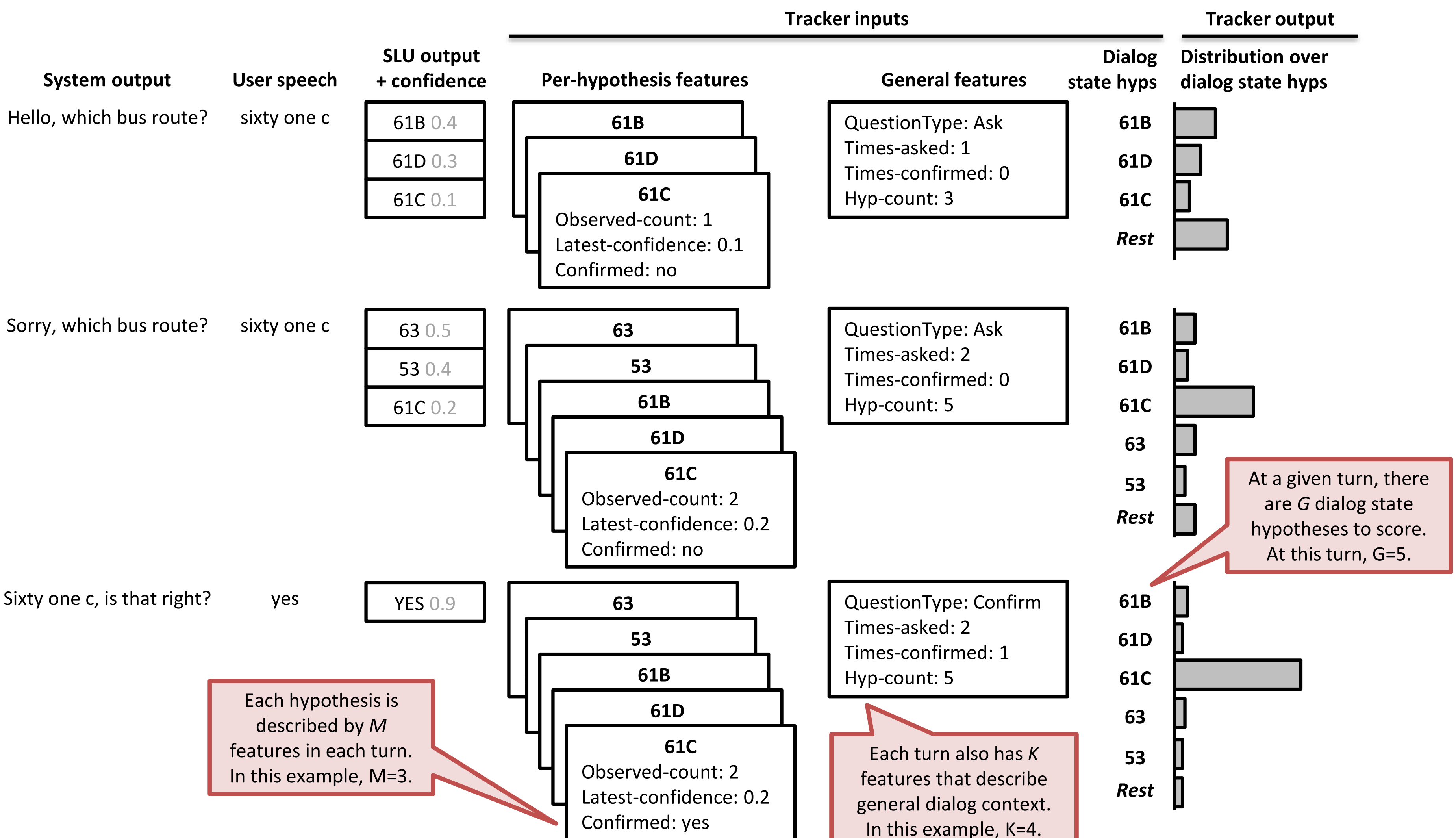
8Khz or 16KHz English ***or French*** ***or Chinese***
speech input

Multilingual ASR Summary Results

Speech Recognizers	WER on ENU
DNN trained with only ENU data	**30.9%**
+FRA, retrain all layers with ENU	30.6%
or +FRA, retrain the top layer with ENU	27.3%
or +FRA+ DEU+ ESP+ITA, retrain top layer	**25.3%**

Dialogue State Tracking Example (Jason Williams)

- Prob [CorrectUserGoals_t | DialogueHistory_{1,2,…,t-1}, UserInfo_{1,2,…,t-1}]



DSN Results for Dialogue State Tracking

- Task: Dialog state tracking (defined in Spoken Dialogue Challenge 2010)
- Strong interactions among features → strength of deep networks
- Can be framed as a multiple binary classification problem
- Baseline: carefully tuned, highly optimized Max Entropy classifier (J. Williams)
- Deep Stacking Nets (slightly tuned) achieve similar accuracy% for all 5 slots:

	State of the Art baseline	**Deep Stacking Networks**
Bus route	58.0%	**58.1%**
Origin location	56.4%	**57.1%**
Destination location	66.5%	65.4%
Date	83.9%	**84.6%**
Time	63.1%	62.5%

Summary

PART I: Basics of Deep Learning (DL)

--- including impact and recent history of DL (Deep Neural Net, DNN) in speech recognition

PART II: Deeper Substance of DL

--- including connections to other ML paradigms
--- two examples of incorporating speech knowledge in DL architectures,
---recent experiments in speech recognition with new DL architectures beyond DNN

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Perspective: What Types of Problems Fit (not fit) Deep Learning (some conjectures)

"Perceptual" AI ⟷ *"Data matching"*

e.g.: Image/video recognition
Speech recognition
Speech/text understanding
Sequential data with temporal structure (stock market prediction?)

e.g.: Malware detection(ICASSP-2013) movie recommender, speaker/language detection?

Non-obvious data representations ⟷ Easy data representation e.g., histogram of events, user-watched movies, etc.

Deep learning already shows tremendous benefits ⟷ Deep learning may not win over standard machine learning

Deep Learning: A Theoretician's Nightmare?

Y LeCun
MA Ranzato

- **Deep Learning involves non-convex loss functions**
  - With non-convex losses, all bets are off
  - Then again, every speech recognition system ever deployed has used non-convex optimization (GMMs are non convex).

- **But to some of us all “interesting” learning is non convex**
  - Convex learning is invariant to the order in which sample are presented (only depends on asymptotic sample frequencies).
  - Human learning isn't like that: we learn simple concepts before complex ones. The order in which we learn things matter.

Deep Learning: A Theoretician's Nightmare?

Y LeCun
MA Ranzato

- **No generalization bounds?**
  - Actually, the usual VC bounds apply: most deep learning systems have a finite VC dimension
  - We don't have tighter bounds than that.
  - But then again, how many bounds are tight enough to be useful for model selection?

- **It's hard to prove anything about deep learning systems**
  - Then again, if we only study models for which we can prove things, we wouldn't have speech, handwriting, and visual object recognition systems today.

Deep Learning: A Theoretician's Paradise?

Y LeCun
MA Ranzato

- **Deep Learning is about representing high-dimensional data**
  - There has to be interesting theoretical questions there
  - What is the geometry of natural signals?
  - Is there an equivalent of statistical learning theory for unsupervised learning?
  - What are good criteria on which to base unsupervised learning?
- **Deep Learning Systems are a form of latent variable factor graph**
  - Internal representations can be viewed as latent variables to be inferred, and deep belief networks are a particular type of latent variable models.
  - The most interesting deep belief nets have intractable loss functions: how do we get around that problem?
- **Lots of theory at the 2012 IPAM summer school on deep learning**
  - Wright's parallel SGD methods, Mallat's "scattering transform", Osher's "split Bregman" methods for sparse modeling, Morton's "algebraic geometry of DBN",....

Deep Learning and Feature Learning Today

Y LeCun
MA Ranzato

- **Deep Learning has been the hottest topic in speech recognition in the last 2 years**
  - A few long-standing performance records were broken with deep learning methods
  - Microsoft and Google have both deployed DL-based speech recognition system in their products
  - Microsoft, Google, IBM, Nuance, AT&T, and all the major academic and industrial players in speech recognition have projects on deep learning
- **Deep Learning is the hottest topic in Computer Vision**
  - Feature engineering is the bread-and-butter of a large portion of the CV community, which creates some resistance to feature learning
  - But the record holders on ImageNet and Semantic Segmentation are convolutional nets
- **Deep Learning is becoming hot in Natural Language Processing**
- **Deep Learning/Feature Learning in Applied Mathematics**
  - The connection with Applied Math is through sparse coding, non-convex optimization, stochastic gradient algorithms, etc...

In Many Fields, Feature Learning Has Caused a Revolution (methods used in commercially deployed systems)

Y LeCun
MA Ranzato

- **Speech Recognition I (late 1980s)**
  - Trained mid-level features with Gaussian mixtures (2-layer classifier)
- **Handwriting Recognition and OCR (late 1980s to mid 1990s)**
  - Supervised convolutional nets operating on pixels
- **Face & People Detection (early 1990s to mid 2000s)**
  - Supervised convolutional nets operating on pixels (YLC 1994, 2004, Garcia 2004)
  - Haar features generation/selection (Viola-Jones 2001)
- **Object Recognition I (mid-to-late 2000s: Ponce, Schmid, Yu, YLC....)**
  - Trainable mid-level features (K-means or sparse coding)
- **Low-Res Object Recognition: road signs, house numbers (early 2010's)**
  - Supervised convolutional net operating on pixels
- **Speech Recognition II (circa 2011)**
  - Deep neural nets for acoustic modeling
- **Object Recognition III, Semantic Labeling (2012, Hinton, YLC,...)**

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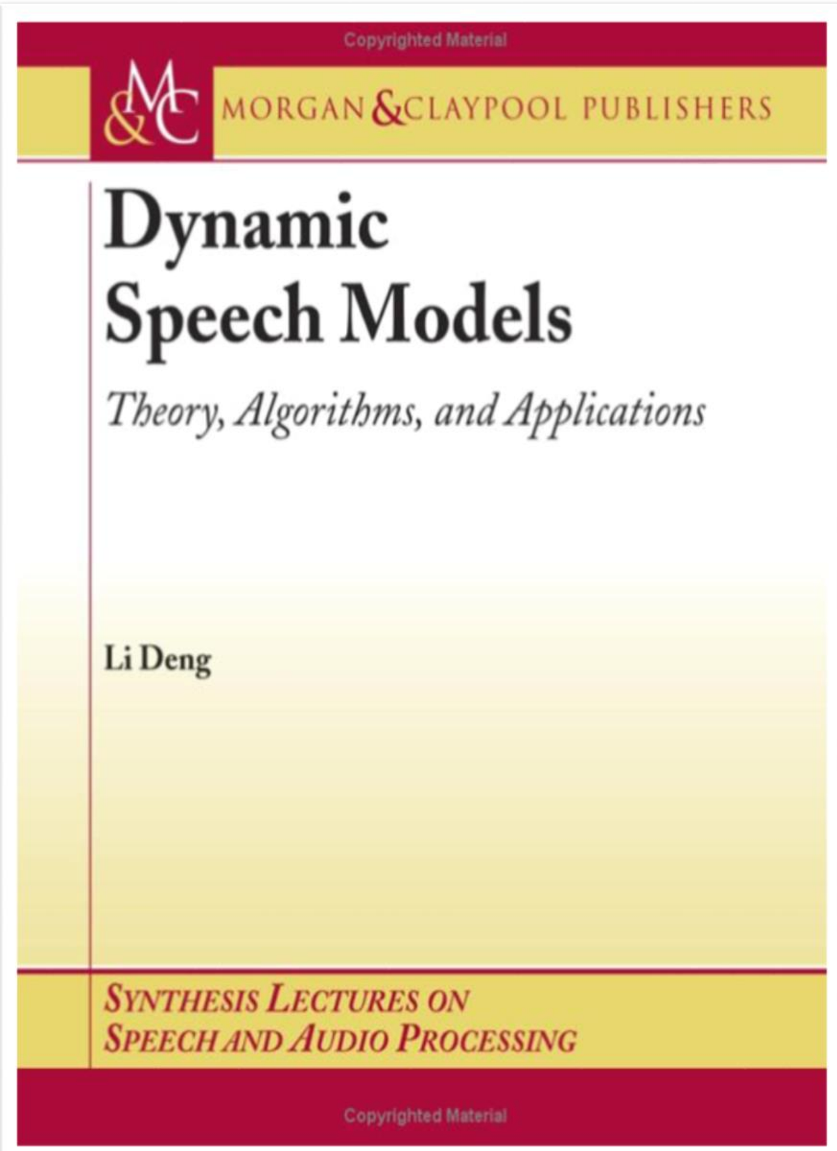

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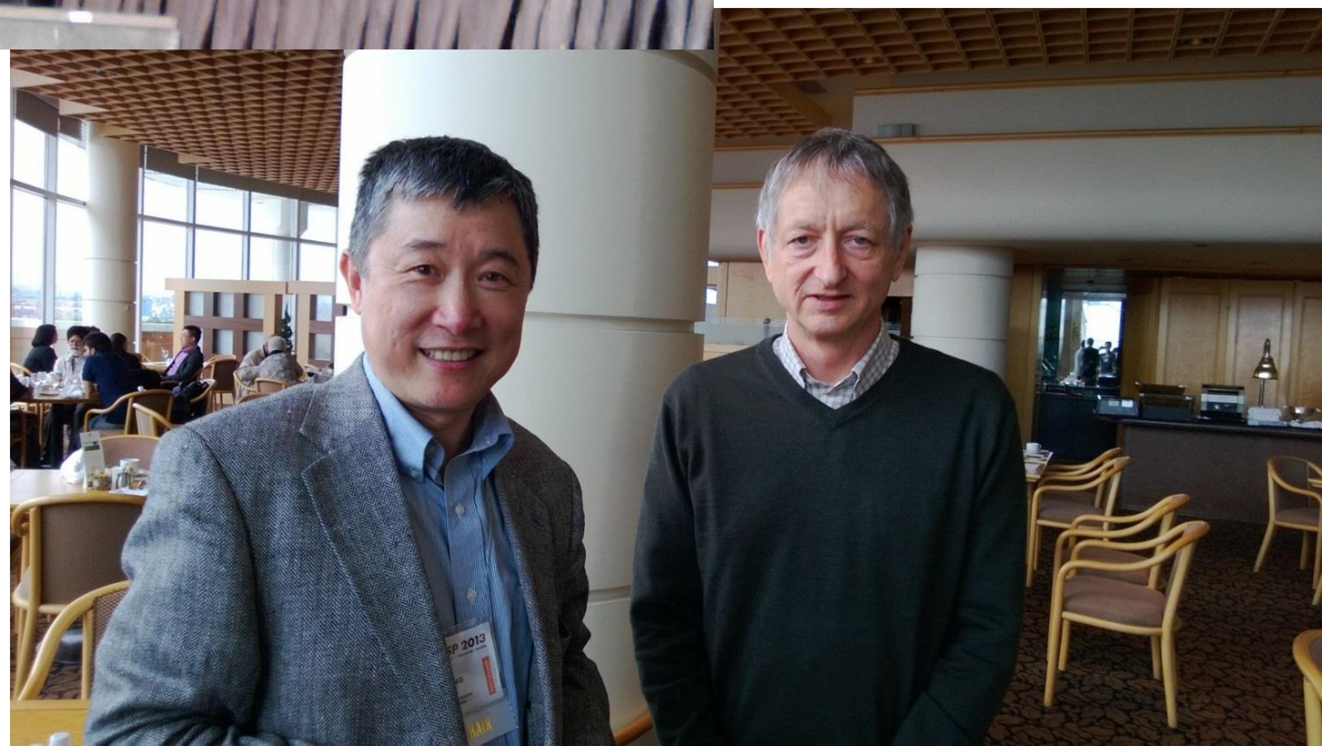
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"DBN vs DBN" (for fun)

From: Geoffrey Hinton [mailto:geoffrey.hinton@gmail.com]
Sent: Tuesday, January 17, 2012 9:33 AM
To: Li Deng
Subject: DBNs are beating DBNs

http://acronyms.thefreedictionary.com/DBN Acronym	Definition
DBN	1, 5-Diazabicyclo(4.3.0)Non-5-Ene (chemical compound)
DBN	Doing Business - Not
DBN	Dialog Broadband Networks (Dialog Telekom PLC; Sri Lanka)
DBN	De Bonis Non (Legal: appointment of a personal representative to a vacancy)
DBN	Divisible by None (band)
DBN	**Deep Belief Network (machine learning)**
DBN	**Dynamic Bayes Network**
DBN	Data Bus Network
DBN	Dial-Back Number
DBN	Day Beacon
DBN	Domain-Border Node
DBN	Digital Billboard Network (Australia)
DBN	Drunk Before Noon
DBN	District Borough Number (New York City Department of Education school identifier)
DBN	Database Notification
DBN	Directed Bipartite Network