

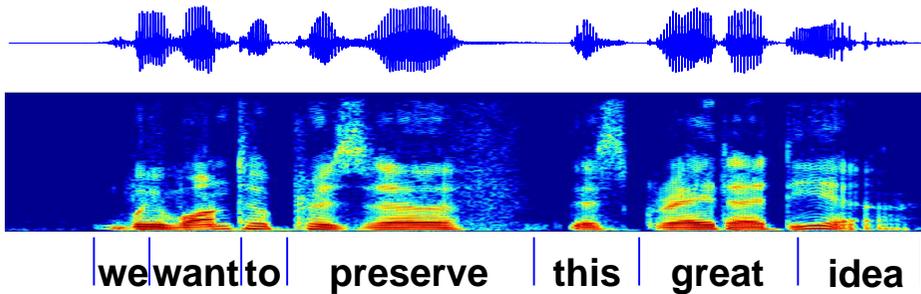
Anniversary Symposium *InterACT* – 25 years
Karlsruhe Institute of Technology (KIT), Baden-Baden, Germany,
July 14/15, 2016

On Architectural Issues of Neural Networks in Speech Recognition

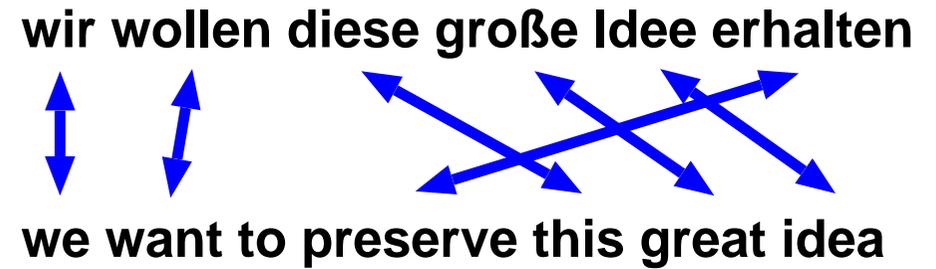
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Automatic Speech Recognition (ASR)



Statistical Machine Translation (SMT)



Handwriting Recognition (Text Image Recognition)



tasks:

- speech recognition
- machine translation
- handwriting recognition

unifying view:

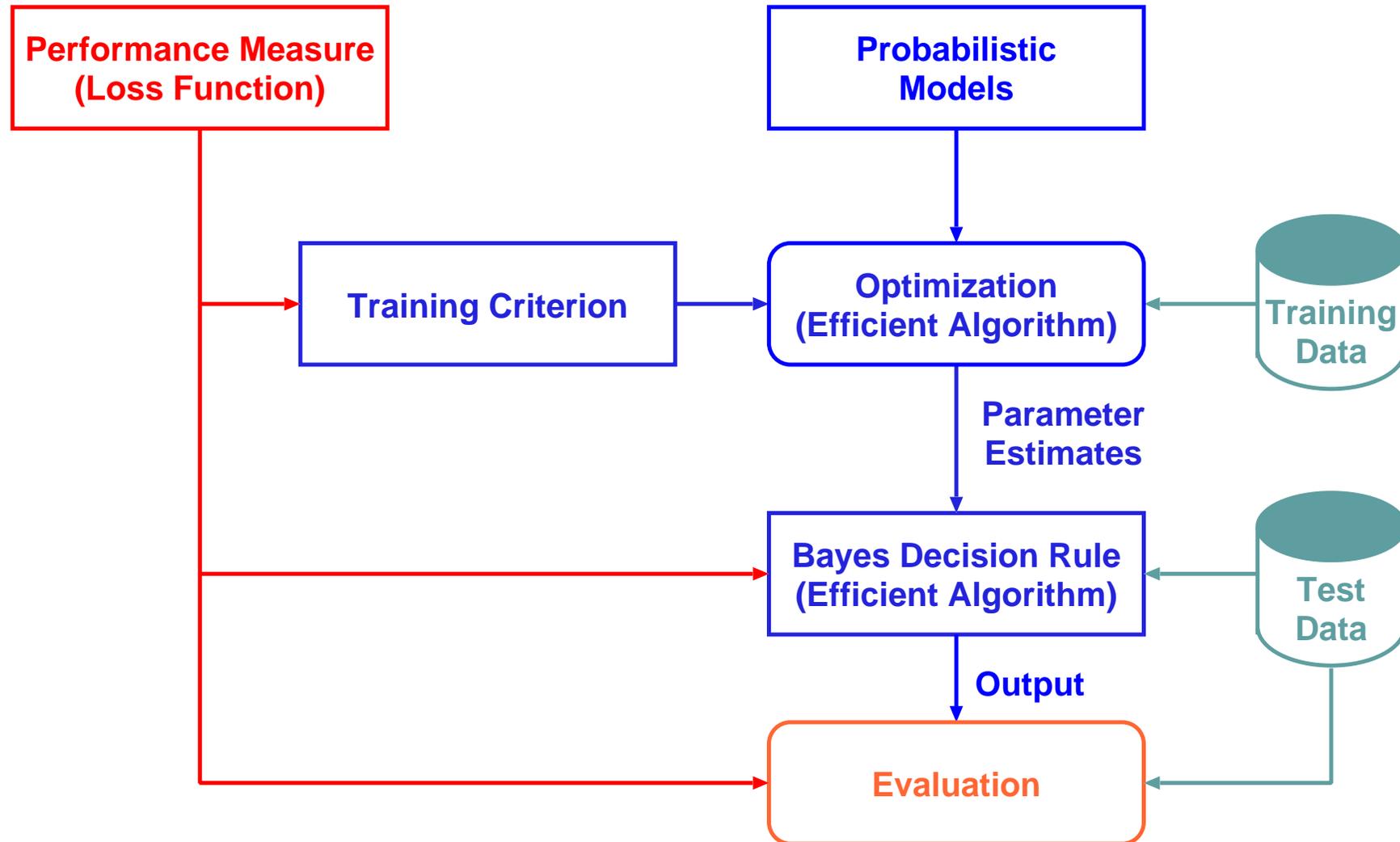
- input string → output string
- output string: natural language

- **VERBMOBIL 1993-2000: funded by German BMBF**
toy task (8000-word vocabulary): recognition and translation for appointment scheduling
- **TC-STAR 2004-2007: funded by EU**
 - real-life task, open domain, large vocabulary:
first research system for speech translation (EU parliament)
 - partners: KIT Karlsruhe, FBK Trento, LIMSI Paris, UPC Barcelona, IBM-US Research, ...
- **GALE 2005-2011: funded by US DARPA**
 - emphasis on Chinese and Arabic speech and text
 - largest project ever on speech and language: 40 Mio USD per year
- **BOLT 2011-2015: funded by US DARPA**
emphasis on colloquial text for Arabic and Chinese
- **QUAERO 2008-2013: funded by OSEO France**
European languages, more colloquial speech, handwriting
- **EU-BRIDGE 2012-2014: funded by EU**
emphasis on recognition and translation of lectures (TED, ...)
- **BABEL 2012-2016: funded by US IARPA**
speech recognition for low-resource languages (and noisy audio!)

evaluations of ASR and SMT systems:

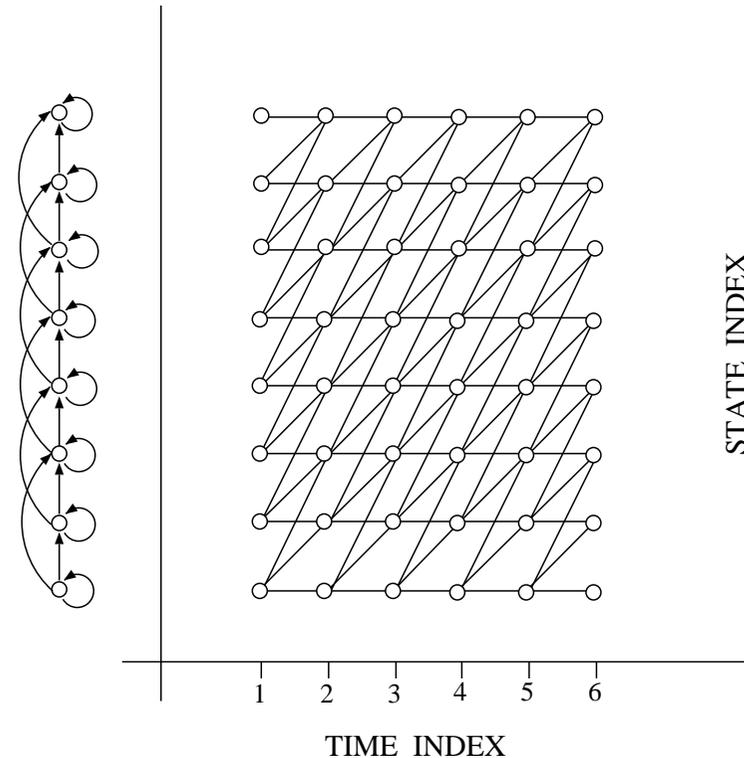
- **project related evaluations:**
 - VERBMOBIL
 - TC-STAR
 - QUAERO
 - EU-BRIDGE
- **public evaluation campaigns:**
 - NIST/LDC/DARPA
 - IWSLT (organized by InterACT members)
 - ACL WMT
- **joint submissions with KIT/InterACT:**
 - system combination

Statistical Approach: No Alternative (incl. Artificial Neural Networks!)



Hidden Markov Models (HMM)

- **fundamental problem in ASR:**
non-linear time alignment
- **Hidden Markov Model:**
 - linear chain of states $s = 1, \dots, S$
 - transitions: forward, loop and skip
- **trellis:**
 - unfold HMM over time $t = 1, \dots, T$
 - path: state sequence $s_1^T = s_1 \dots s_t \dots s_T$
 - observations: $x_1^T = x_1 \dots x_t \dots x_T$



general view:

- two sequences without synchronization: acoustic vectors and states (with labels)
- HMM: mechanism that takes care of the synchronization (=alignment) problem

Hidden Markov Models (HMM)

The acoustic model $p(X|W)$ provides the link between word sequence hypothesis W and observations sequence $X = x_1^T = x_1 \dots x_t \dots x_T$:

- acoustic probability $p(x_1^T|W)$ using hidden state sequences s_1^T :

$$p(x_1^T|W) = \sum_{s_1^T} p(x_1^T, s_1^T|W) = \sum_{s_1^T} \prod_t [p(s_t|s_{t-1}, W) \cdot p(x_t|s_t, W)]$$

- two types of distributions:
 - transition probability $p(s|s', W)$: not important
 - emission probability $p(x_t|s, W)$: key quantity
realized by GMM: Gaussian mixtures models (trained by EM algorithm)
- phonetic labels (allophones, sub-phones): $(s, W) \rightarrow a = a_{sW}$

$$p(x_t|s, W) = p(x_t|a_{sW})$$

typical approach: phoneme models in triphone context:
decision trees (CART) for finding equivalence classes

- refinements:
 - augmented feature vector: context window around position t
 - subsequent LDA (linear discriminant analysis)

Hybrid Approach: HMM and ANN

consider modelling the acoustic vector x_t in an HMM:

- re-write the emission probability for annotation label a and acoustic vector x_t (strictly speaking: an approximation only):

$$p(x_t|a) = p(x_t) \cdot \frac{p(a|x_t)}{p(a)}$$

- prior probability $p(a)$: estimated as relative frequencies
- for recognition purposes: the term $p(x_t)$ can be dropped
- result: model the label posterior probability by an ANN:

$$x_t \rightarrow p(a|x_t)$$

rather than the state emission distribution $p(x_t|a)$

- justification:
 - easier learning problem: labels $a = 1, \dots, 5000$ vs. vectors $x_t \in \mathbb{R}^{D=40}$
 - well-known result in pattern recognition/machine learning;
but ignored in ASR due to the mathematical beauty of the EM algorithm

History: ANN in Acoustic Modelling

- 1988 [Waibel & Hanazawa⁺ 88]:
phoneme recognition using time-delay neural networks
- 1989 [Bridle 89]:
softmax operation for probability normalization in output layer
- 1990 [Bourlard & Wellekens 90]:
 - for squared error criterion, ANN outputs can be interpreted as class posterior probabilities (rediscovered: Patterson & Womack 1966)
 - they advocated the *hybrid approach*: use the ANN outputs to replace the emission probabilities in HMMs
- 1993 [Haffner 93]:
sum over label-sequence posterior probabilities in hybrid HMMs
- 1994 [Robinson 94]: recurrent neural network
 - competitive results on WSJ task
 - his work remained a singularity in ASR

experimental situation:

- until 2011: ANNs were never really competitive with Gaussian mixture models
- after 2011: yes, *deep learning* [Deng & Hinton 2012]

History: ANN in Acoustic Modelling

more ANN approaches:

- 1994 [LeCun & Bengio⁺ 94]:
convolutional neural networks
- 1997 A. Waibel's team [Fritsch & Finke⁺ 97]:
hierarchical mixtures of experts
- 1997 [Hochreiter & Schmidhuber 97]:
long short-term memory neural computation
with extensions [Gers & Schraudolph⁺ 02]

renaissance of ANN: concepts of deep learning and related ideas:

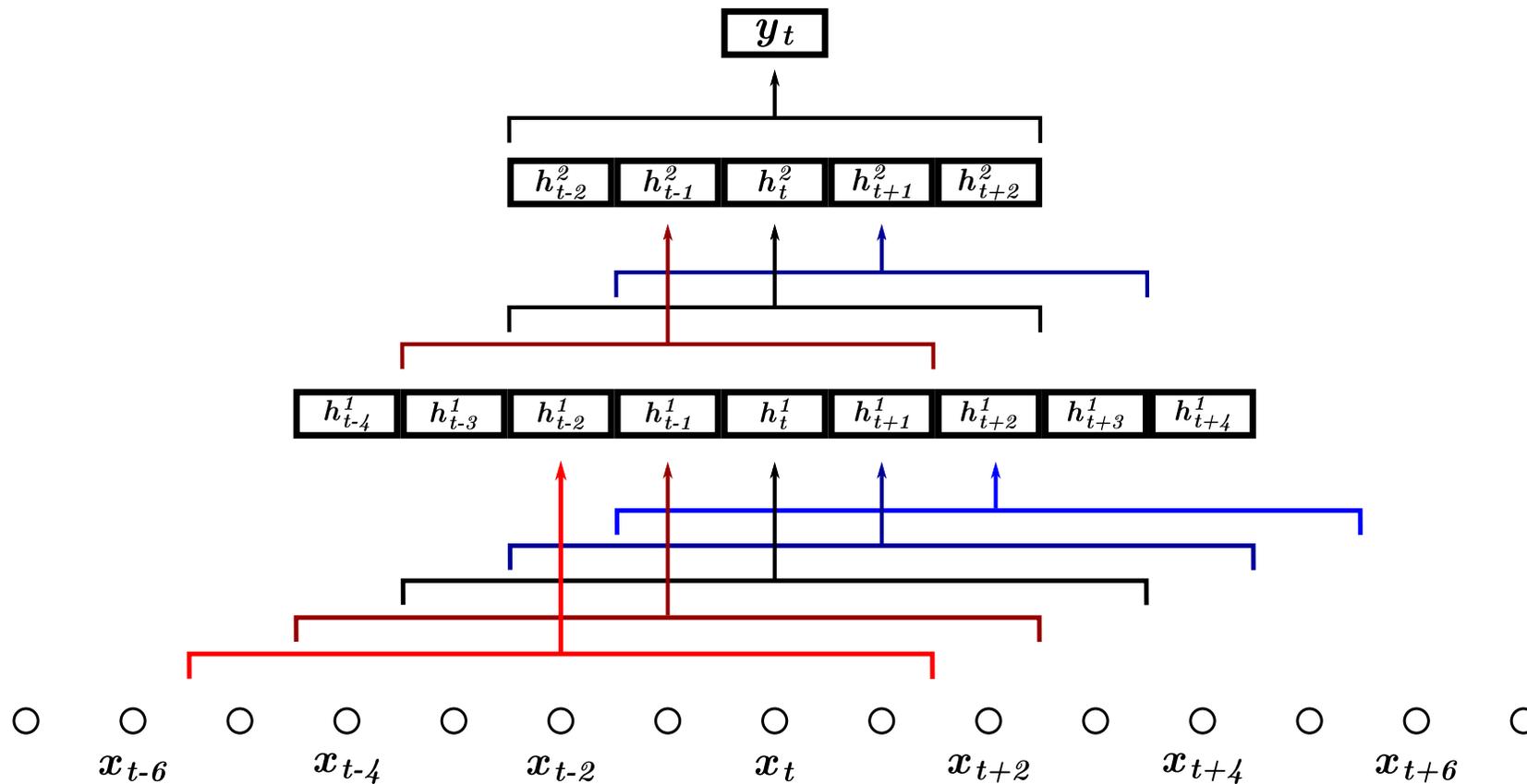
- 2000 [Hermansky & Ellis⁺ 00]: tandem approach: multiple layers of processing
by combining Gaussian model and ANN for ASR
- 2002 [Utgoff & Stracuzzi 02]: many-layered learning for symbolic processing
- 2006 [Hinton & Osindero⁺ 06]: introduced what he called *deep learning (belief nets)*
- 2008 [Graves 08]: good results on LSTM RNN for handwriting task
- 2012 Microsoft Research [Dahl & Yu⁺ 12]:
 - combined Hinton's deep learning with hybrid approach
 - significant improvement by deep MLP on a large-scale task
- since 2012: other teams confirmed significant reductions of WER



TDNN: Time Delay Neural Network

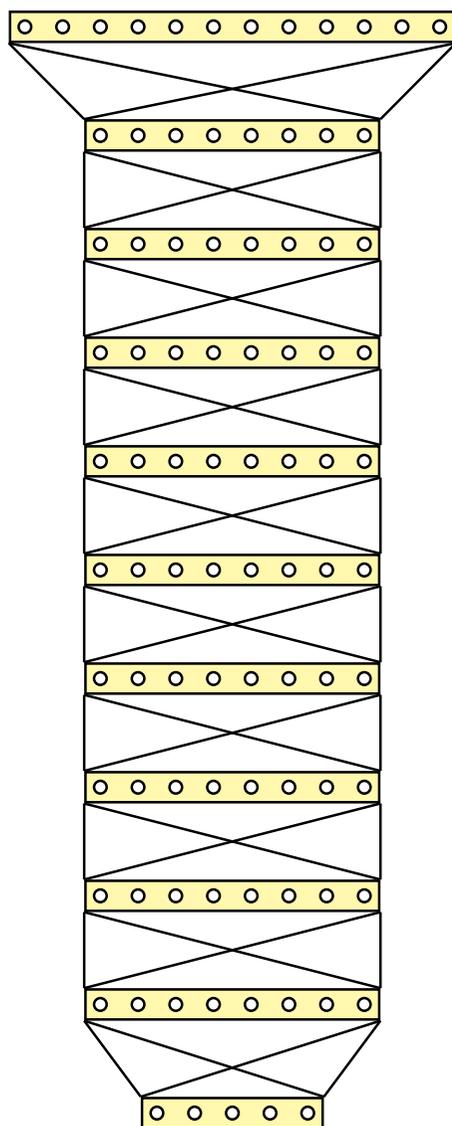
[Waibel & Hanazawa⁺ 88]

- TDNN: feed-forward multi-layer perceptron with special properties:
- long temporal context
 - weight sharing



- first (?) publication: [Waibel & Hanazawa⁺ 88] at ICASSP 1988, New York
- full journal paper: [Waibel & Hanazawa⁺ 89] in *IEEE Transactions on Acoustics, Speech and Signal Processing 1989*
 - 2036 citations (*Google Scholar*)
 - 1116 citations on 3 more papers on TDNN 1989/90
- recent work by Dan Povey's team [Peddinti & Povey⁺ 15] at Interspeech 2015: improvements over widely used deep MLP approach
 - on many of the standard ASR tasks (WSJ, Switchboard, Librispeech, ...)
 - on ASPIRE challenge (IARPA, March 2015):
reverberant speech in farfield speech recognition

Today vs. 1988-94: What is Different?



most popular and widely used:

feed-forward multi-layer perceptron (FF MLP)

- operations: matrix · vector
- nonlinear activation function

comparison for ASR: today vs. 1988-1994:

- number of hidden layers:
10 (or more) rather than 2-3
- number of output nodes (phonetic labels):
5000 rather than 50
- optimization strategy:
practical experience and heuristics,
e.g. layer-by-layer pretraining
- much more computing power

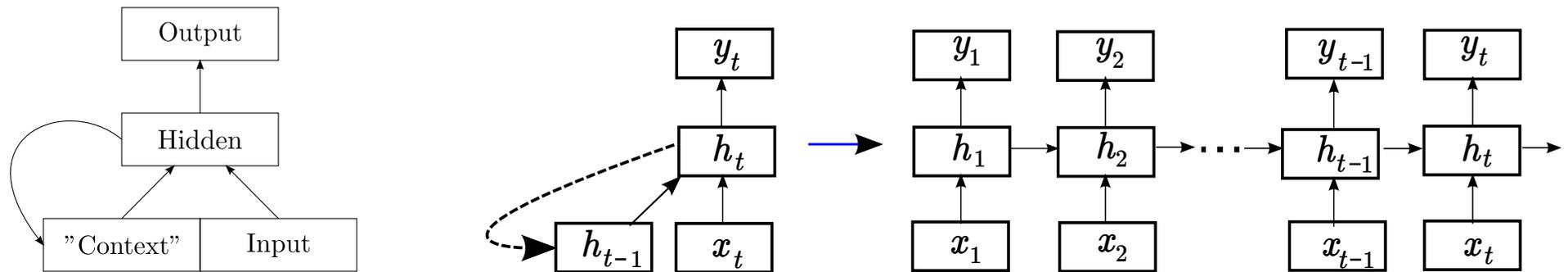
overall result:

- huge improvement by ANN
- WER is (nearly) halved !!

Recurrent Neural Network: String Processing

principle for string processing over time $t = 1, \dots, T$:

- introduce a memory (or context) component to keep track of history
- quantities: input = observation x_t , memory h_{t-1} , output distribution y_t



extensions:

- **bidirectional variant [Schuster & Paliwal 1997]**
- **feedback of output labels**
- **long short-term memory [Hochreiter & Schmidhuber 97; Gers & Schraudolph⁺ 02]**
- **deep structure: several hidden layers**

Direct Model of Label Sequence (spirit of CTC: connectionist temporal classification)

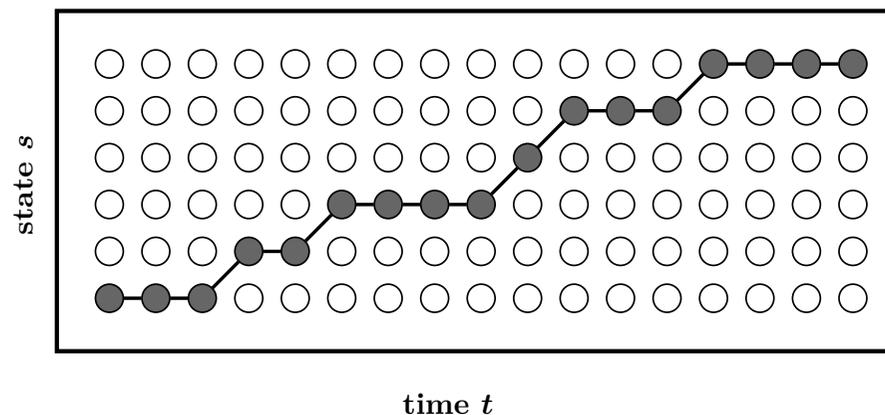
re-formulate the problem of speech recognition:

- sequence of phonetic labels (e.g. CART): $a_s, s = 1, \dots, S$
(which fully determines the sequence of words)
- key quantity: (local) label posterior probability calculated by an ANN

$$p_t(a|x_1^T) = p_t(a|x_{t-\delta}^{t+\delta})$$

- model localization effect by alignments, i.e. mappings from time to states:

$$t \rightarrow s = s_t$$



Direct Model of Label Sequence

sum over all hidden alignments s_1^T :

$$\begin{aligned} p(a_1^S | x_1^T) &= \sum_{s_1^T} p(a_1^S, s_1^T | x_1^T) = \dots \\ &= \sum_{s_1^T} \prod_t p_t(a_{s_t} | x_{t-\delta}^{t+\delta}) \end{aligned}$$

open issues:

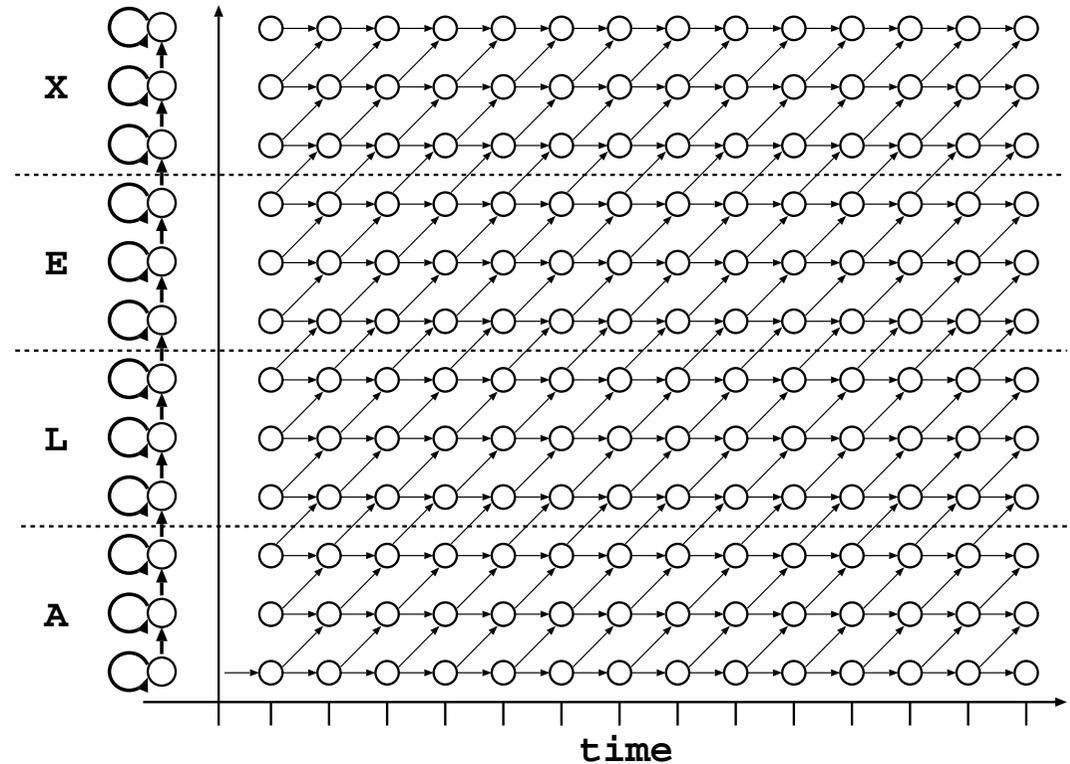
- how to include the transition probabilities
- how to include the language model
- how to perform end-to-end training

requirement:

avoid the global re-normalization as in discriminative/hybrid HMM

Comparison with Discriminative/Hybrid HMM

- topology:
conventional HMM structure
- important differences:
 - no joint model $p(a_1^S, x_1^T)$
 - no global re-normalization (e. g. lattice)
- open issues:
 - transition probabilities
 - language model
 - consistent training criterion:
sum over all alignments,
end-to-end training,...

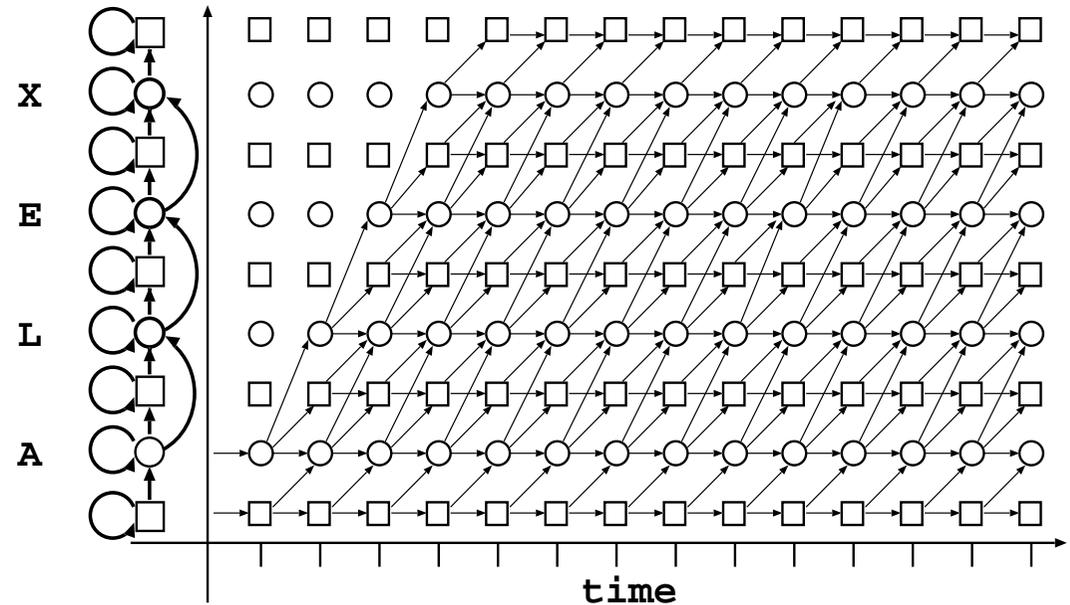


goal: avoid joint probability $p(a_1^S, x_1^T)$ as in discriminative/hybrid HMM

Comparison with CTC: connectionist temporal classification [Graves & Fernandez⁺ 06]

characteristic properties of CTC:

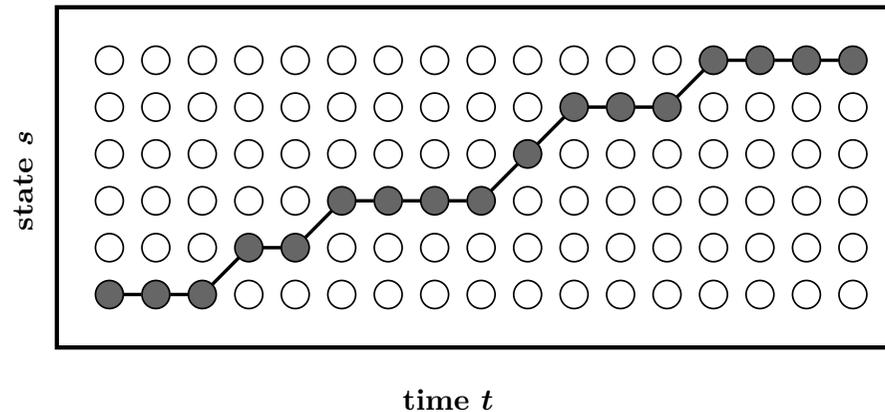
- topology: for each symbol label: single state + blank state
- no transition probabilities
- training criterion: sum
- ANN structure: LSTM RNN or ...?



experiments for CTC and related neural network approaches:

- good results reported
- reason: LSTM RNN?
- direct comparison: to be done

Direct Model of Label Sequence: Inverted Alignments



- re-interpretation of ASR: segmentation and classification problem
- consider inverted alignments, i.e. from state s to time t :

$$s \rightarrow t = t_s$$
- sum over inverted alignments as hidden variables t_1^S :

$$\begin{aligned}
 p(a_1^S | x_1^T) &= \sum_{t_1^S} p(a_1^S, t_1^S | x_1^T) = \dots = \\
 &= \sum_{t_1^S} \prod_{s=1}^S p_{t_s}(a_s | x_1^T) = \sum_{t_1^S} \prod_{s=1}^S p_{t_s}(a_s | x_{t_s-\delta}^{t_s+\delta})
 \end{aligned}$$

experiments: underway

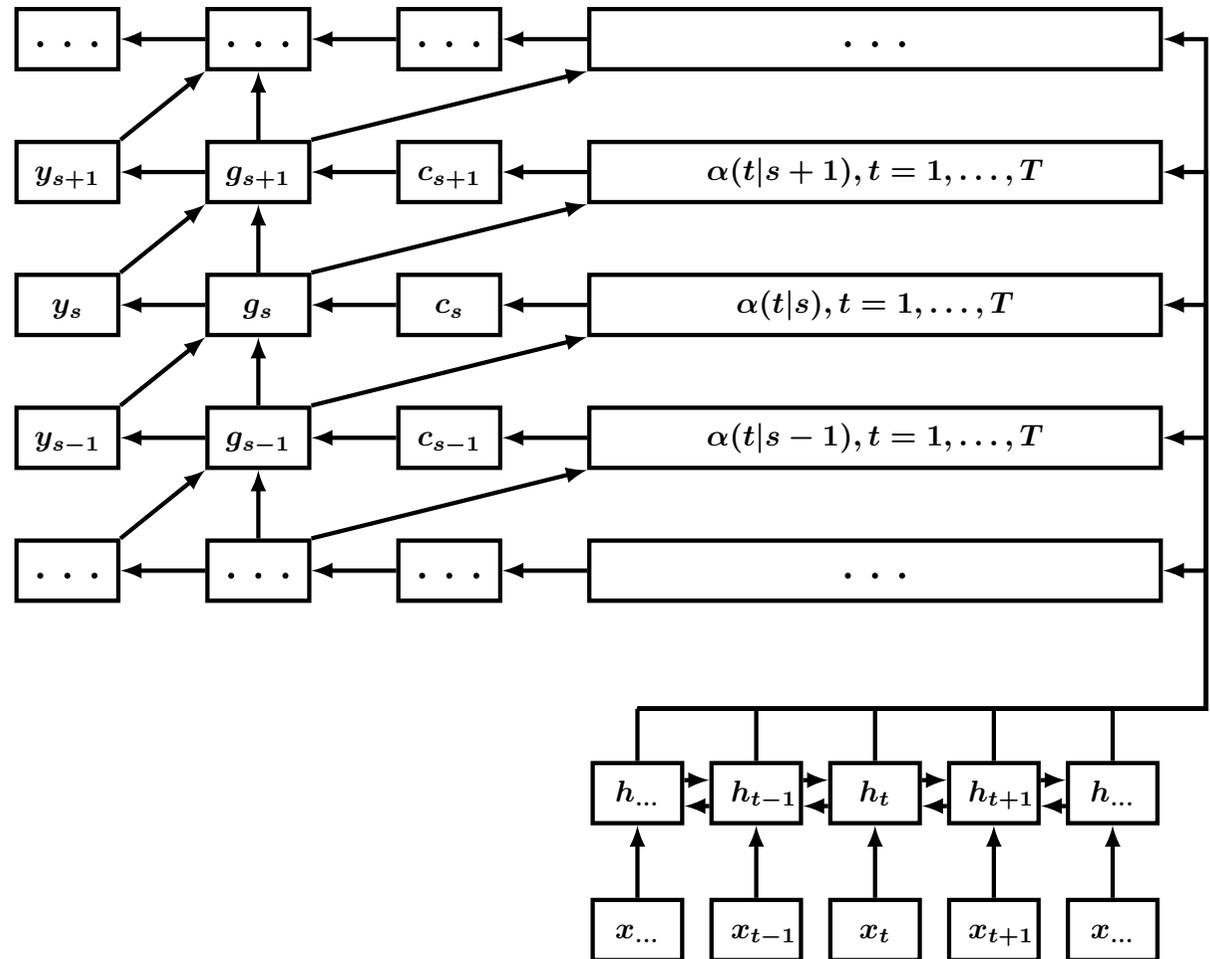
Mechanism of Attention: Alignment by ANN (originally introduced for MT [Bahdanau & Cho⁺ 15])

mechanism of attention:
ANN only

alignment direction:
from state s to time t

occupation probabilities:
 $\alpha(t|s)$

experiments:
ongoing work, many teams



Architectural Issues of ANN in ASR Systems:

- **starting point: direct model of label sequences:**
 - use ANN output as label posterior probability
 - (try to) avoid global re-normalization (no denominator/lattice)
- **open questions:**
 - how to include transition probabilities?
 - how to include language model?
 - end-to-end training: suitable training criterion
- **some localization is needed: alignments**
 - inverted alignments vs. traditional alignments
 - attention-based mechanism: alternative?
- **experimental results: room for improvements**
 - a large number of ongoing studies
 - clear conclusions: difficult

Congratulations to InterACT and Alex on 25 successful years!

Best wishes for the coming 25 years!

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