

End-to-End ASR

Presented by Aku Rouhe

Isn't all ASR end-to-end?

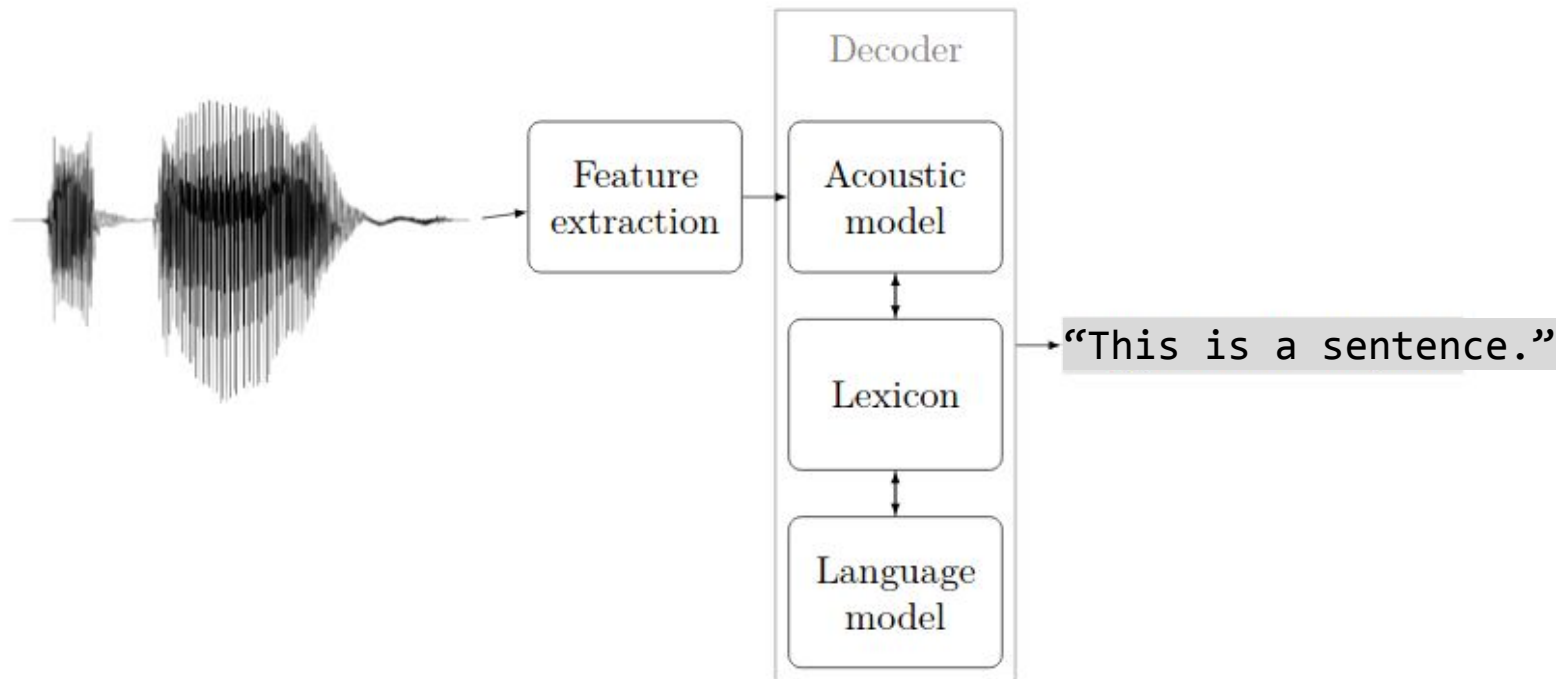


“This is a sentence.”

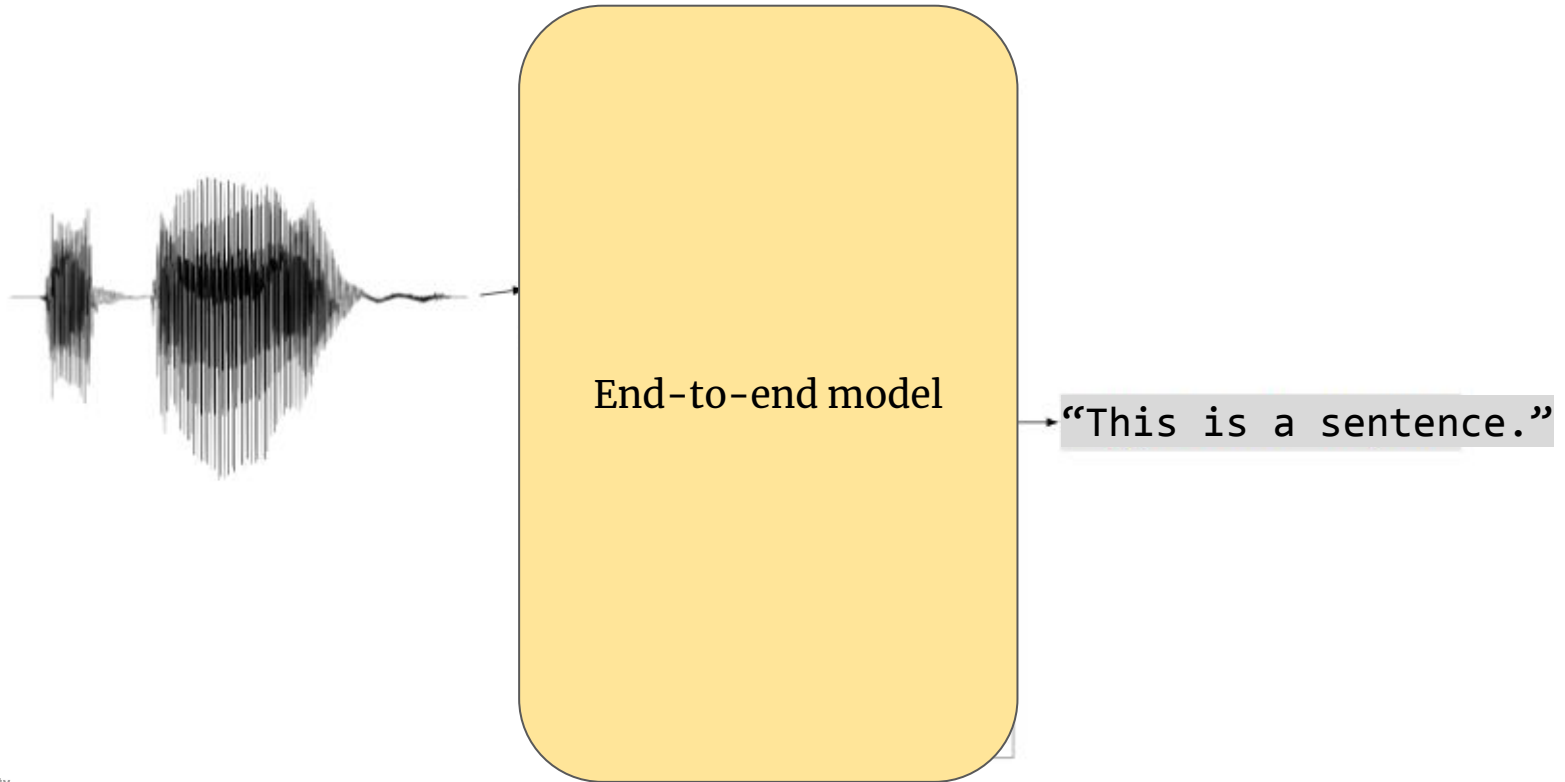
End-to-End is a Vague Umbrella term



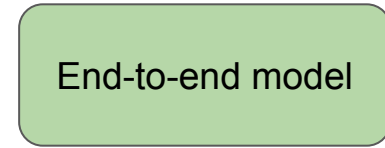
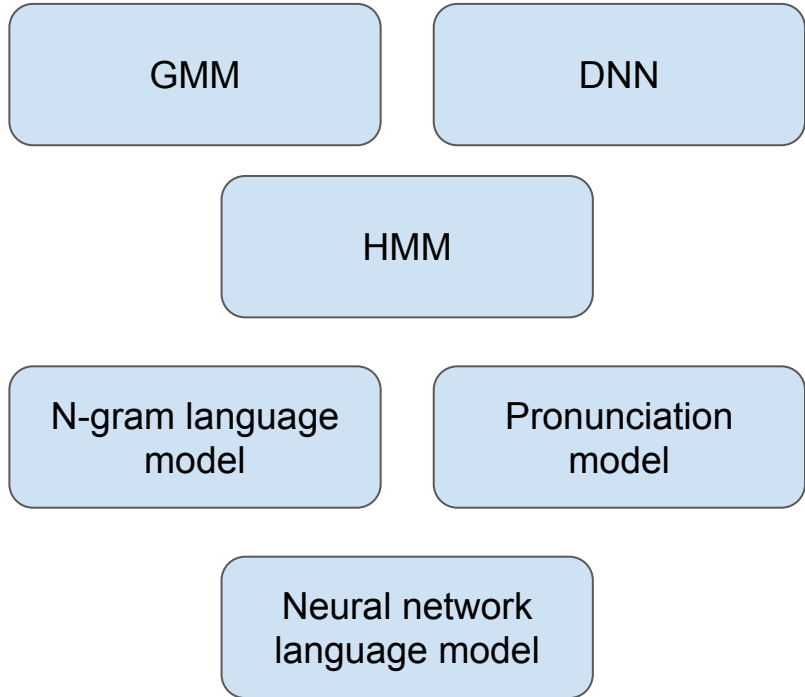
HMM-system: Multiple models



E2E-model: Directly from audio to text



Simplify ASR



A look at search spaces

Multimodel: $\arg_{\mathbf{w}} \max p(\mathbf{0} \mid \mathbf{s})p(\mathbf{s} \mid \mathbf{w})p(\mathbf{w})$

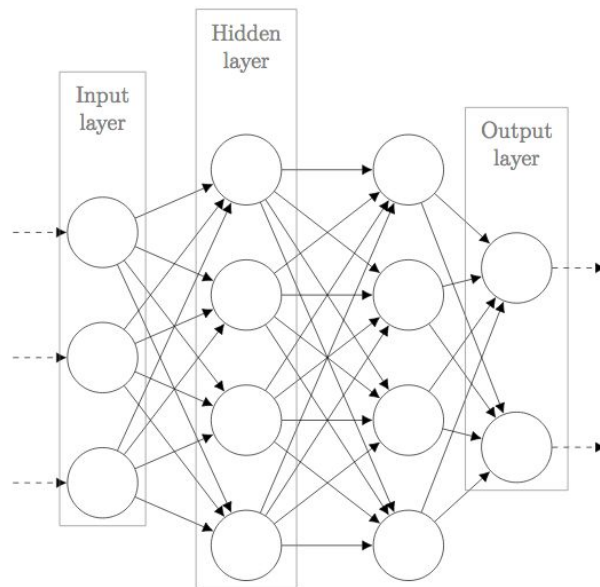
End-to-End: $\arg_{\mathbf{w}} \max p(\mathbf{w} \mid \mathbf{0})$

Joint training, Joint decoding

- *Joint decoding*: Use all submodels together - before pruning
 - e.g. Decoding algorithm combines $p(\mathbf{o} \mid \mathbf{s})$, $p(\mathbf{s} \mid \mathbf{w})$, and $p(\mathbf{w})$
- *Joint training*: Train all submodels together - avoid suboptimization
 - e.g. One global training criterion

How to model $p(\mathbf{w} | \mathbf{o})$ directly?

Use a big neural network



Is End-to-End better?

- Not necessarily in terms of WER
- End-to-End systems can more easily run on e.g. a mobile phone

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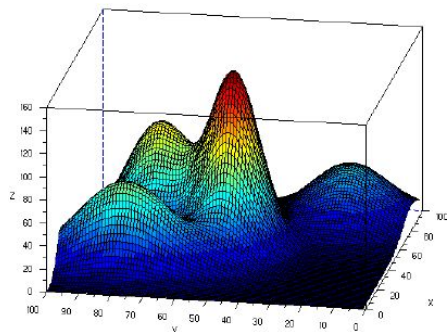
- Connectionist Temporal Classification
- Neural Transducer
- BREAK
- Attention-based Encoder-Decoder

Kahoot

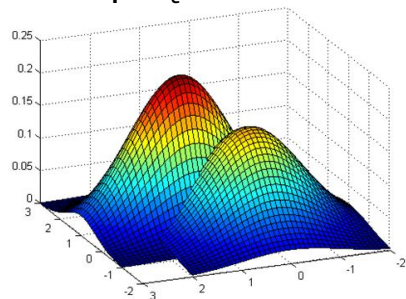
Background from HMM Acoustic Models

GMM

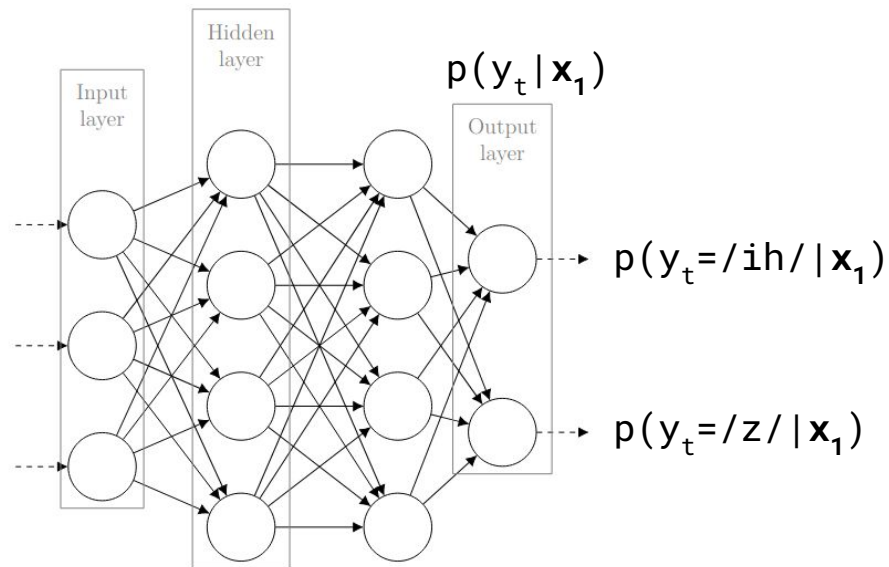
$$p(\mathbf{x}_1 | y_t = /ih/)$$



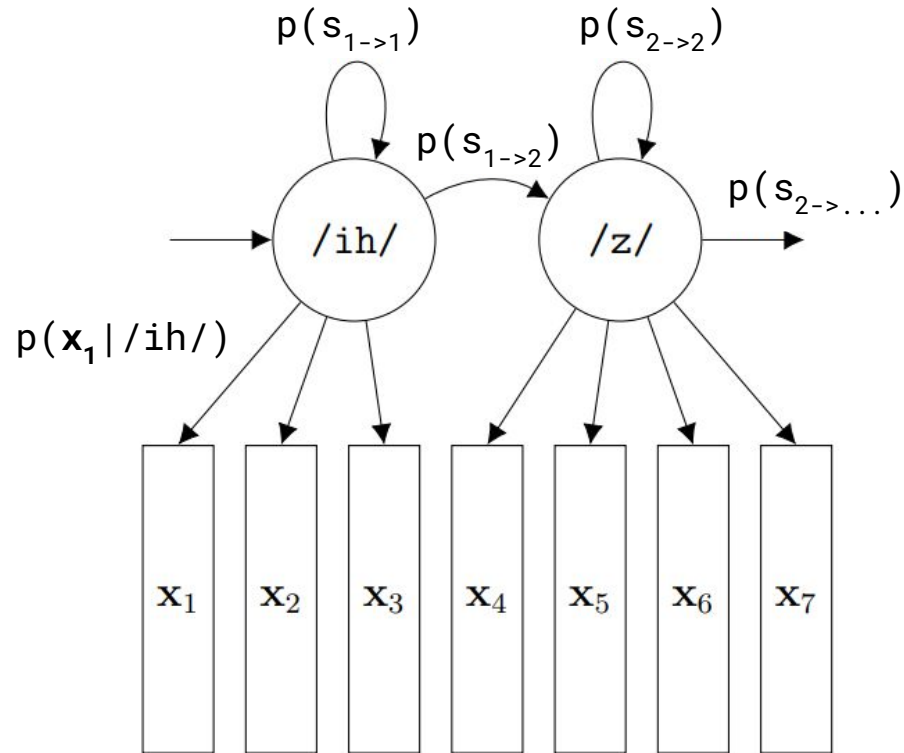
$$p(\mathbf{x}_1 | y_t = /z/)$$



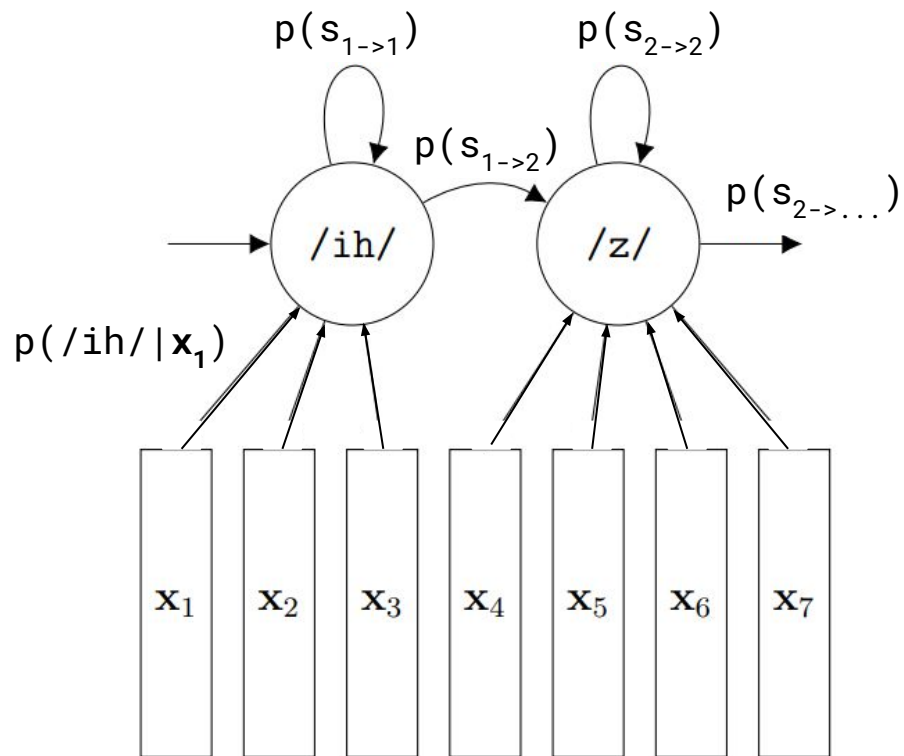
DNN



HMM / GMM

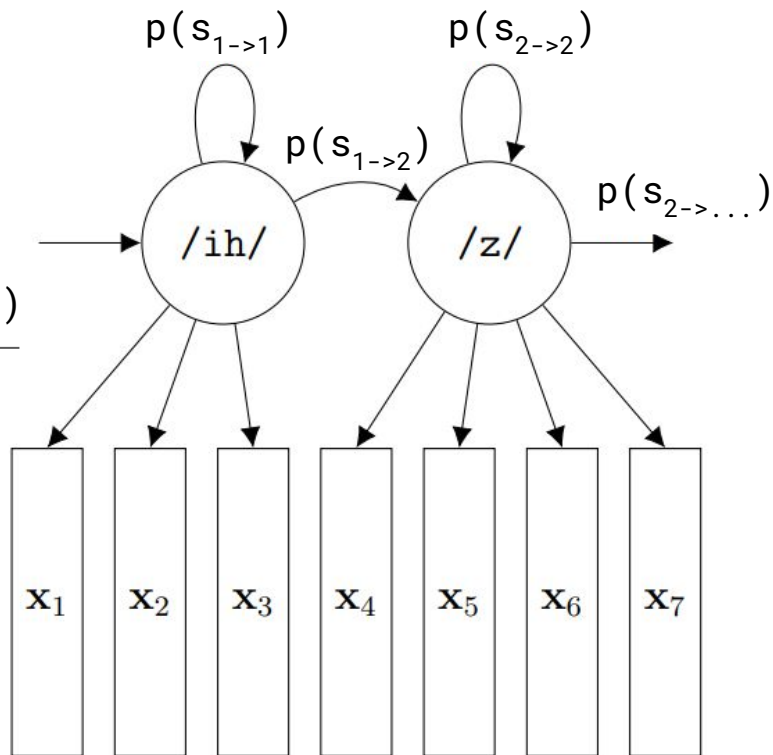


HMM / DNN

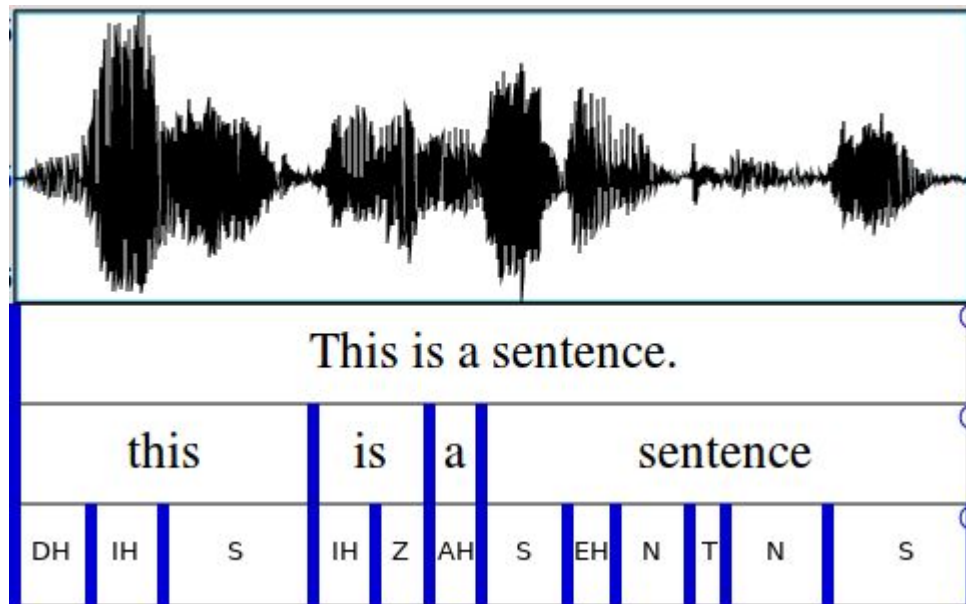
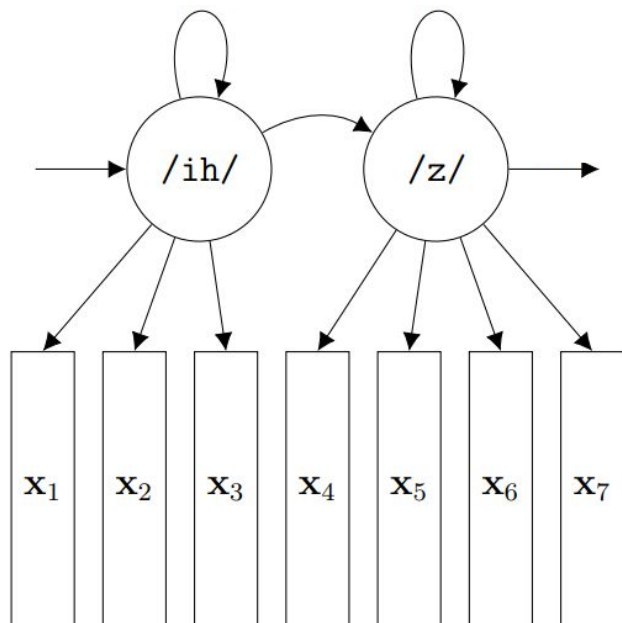


HMM / DNN

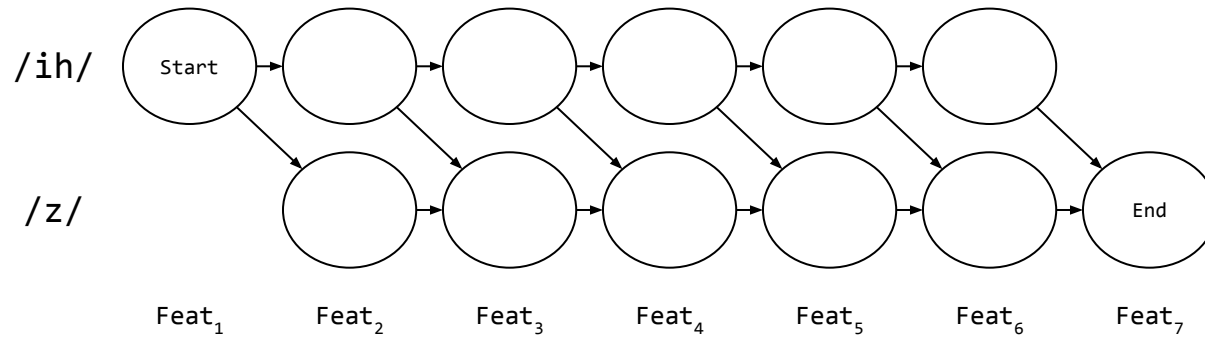
$$\frac{p(\mathbf{x}_1 | /ih/)}{p(\mathbf{x}_1)} = \frac{p(/ih/ | \mathbf{x}_1)}{p(/ih/)}$$



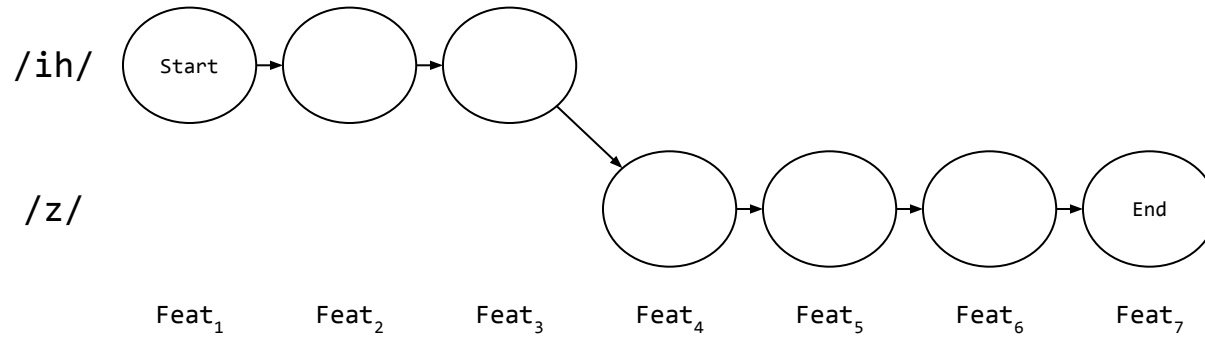
HMM Alignment



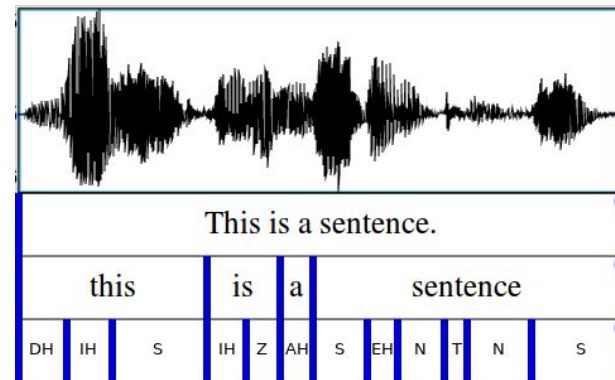
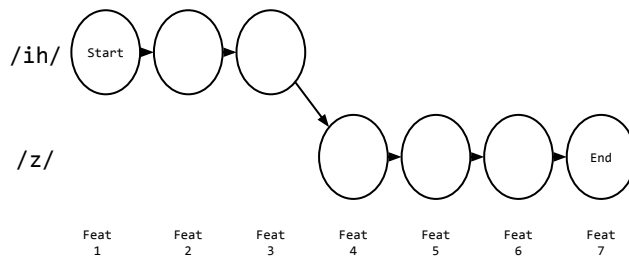
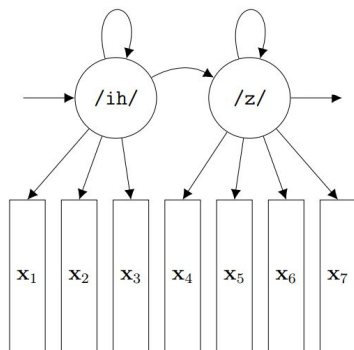
Full-Sum Training (Forward-Backward)



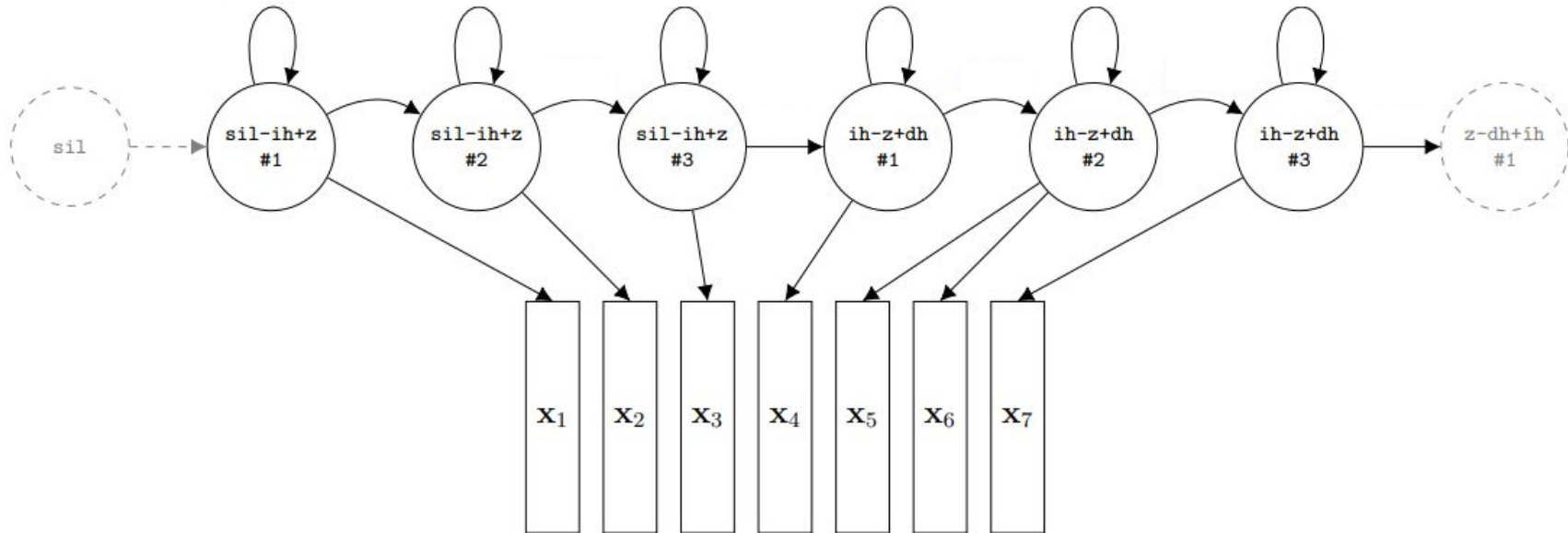
Viterbi



HMM Alignment

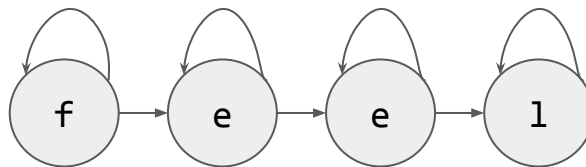
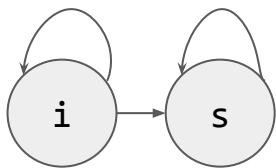


Triphone Tristate HMM



A simpler HMM / DNN system?

- Full sum training doesn't need existing alignments
- What about tristate triphone HMMs and the state tying they need - could we do without it?
- What about phone units - could do without them as well, and just use characters?



Connectionist Temporal Classification

*Connectionist temporal classification:
labelling unsegmented sequence data
with recurrent neural networks*

Alex Graves, Santiago Fernández,
Faustino Gomez, and Jürgen
Schmidhuber

2006

In Proceedings of the 23rd
international conference on Machine
learning (ICML)

CTC output

h h e ϵ ϵ l l l ϵ l l o

h e ϵ l ϵ l o

h e l l o

h e l l o

First, merge repeat characters.

Then, remove any ϵ tokens.

The remaining characters are the output.

© Awni Hannun, Distill

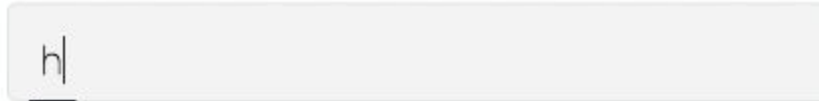
Connectionist Temporal Classification (CTC)

How CTC collapsing works

For an input,
like speech



Predict a
sequence of
tokens



Use `<return>` to
input a blank (ϵ)

Merge repeats,
drop ϵ



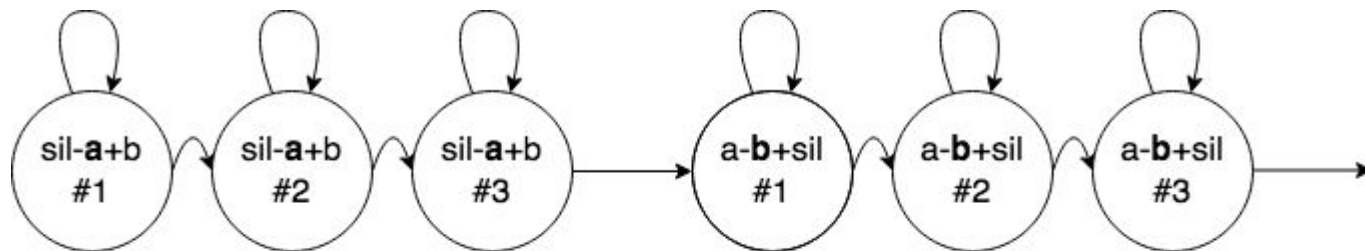
Final output



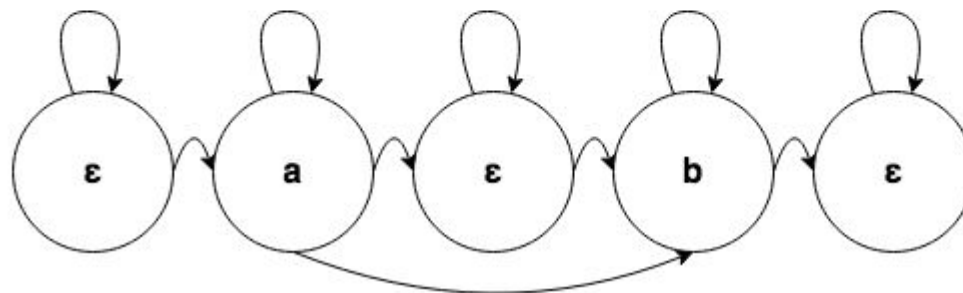
<https://distill.pub/2017/ctc/>

CTC Graph

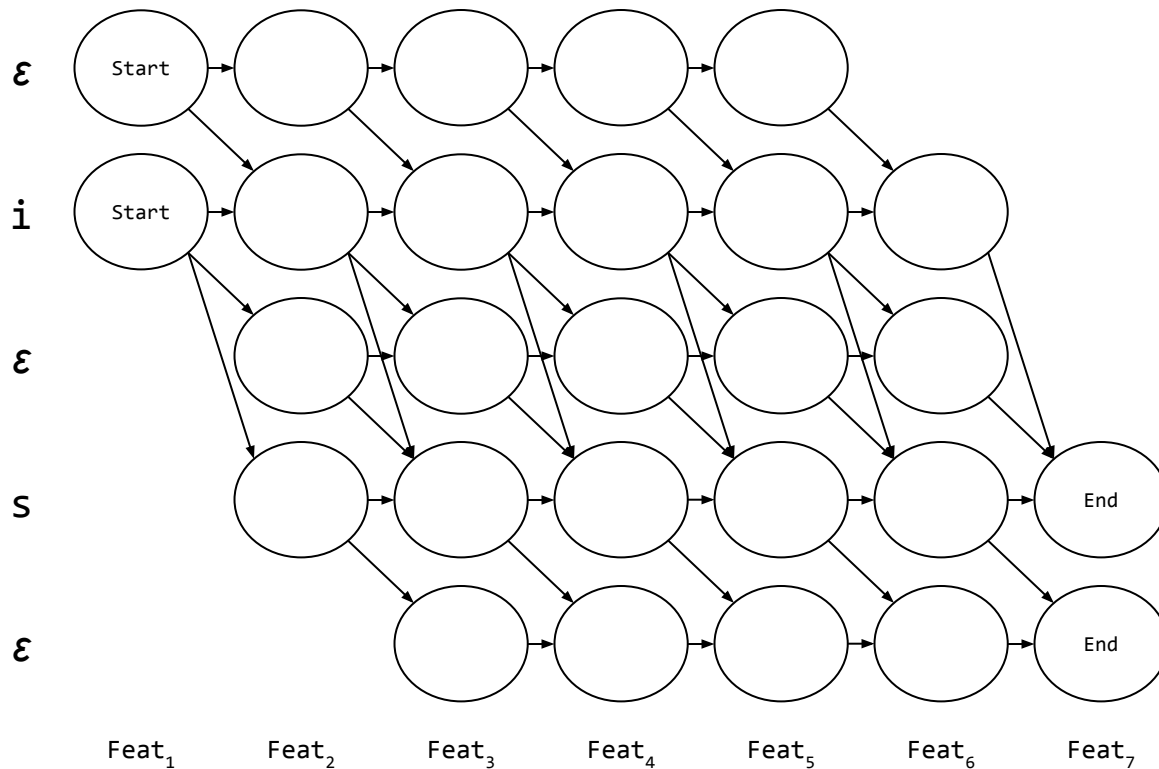
Linear
HMM

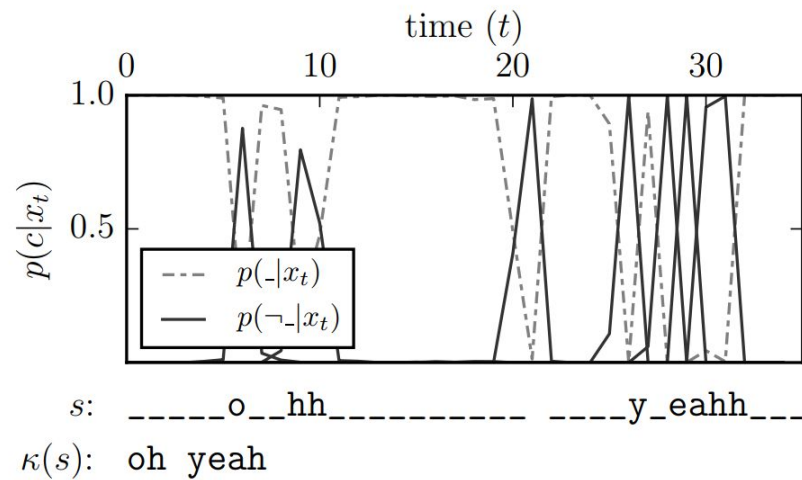


CTC

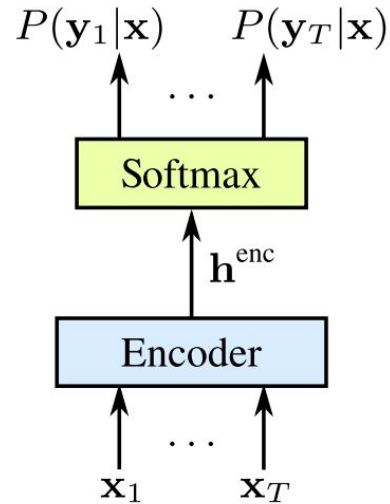


CTC Full-Sum Training





Connectionist Temporal Classification



Conditional independence assumption in CTC

$$P(Y_t \mid X_{1..t})$$

Neural Transducer

*Sequence Transduction with Recurrent
Neural Networks*

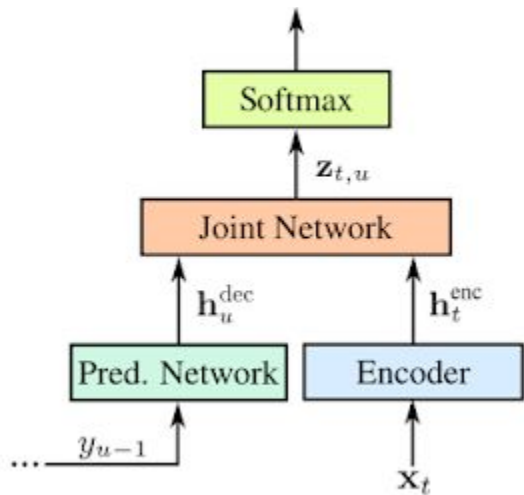
Alex Graves

2012

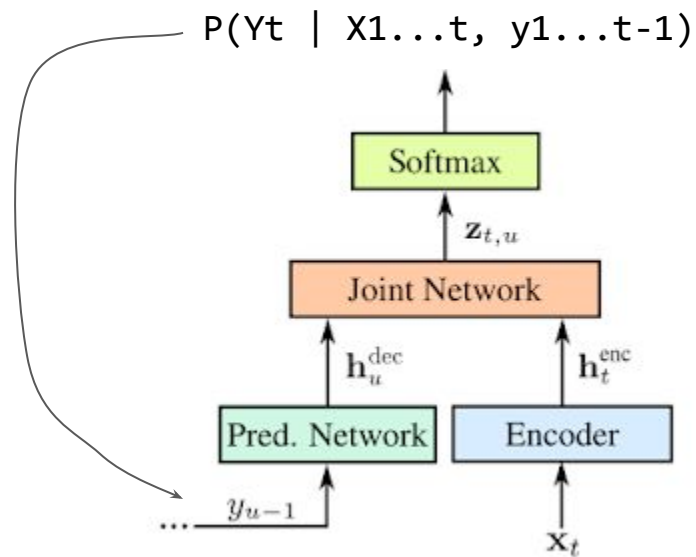
In ICML Workshop on Representation
Learning

Neural Transducer (sometimes RNN-Transducer)

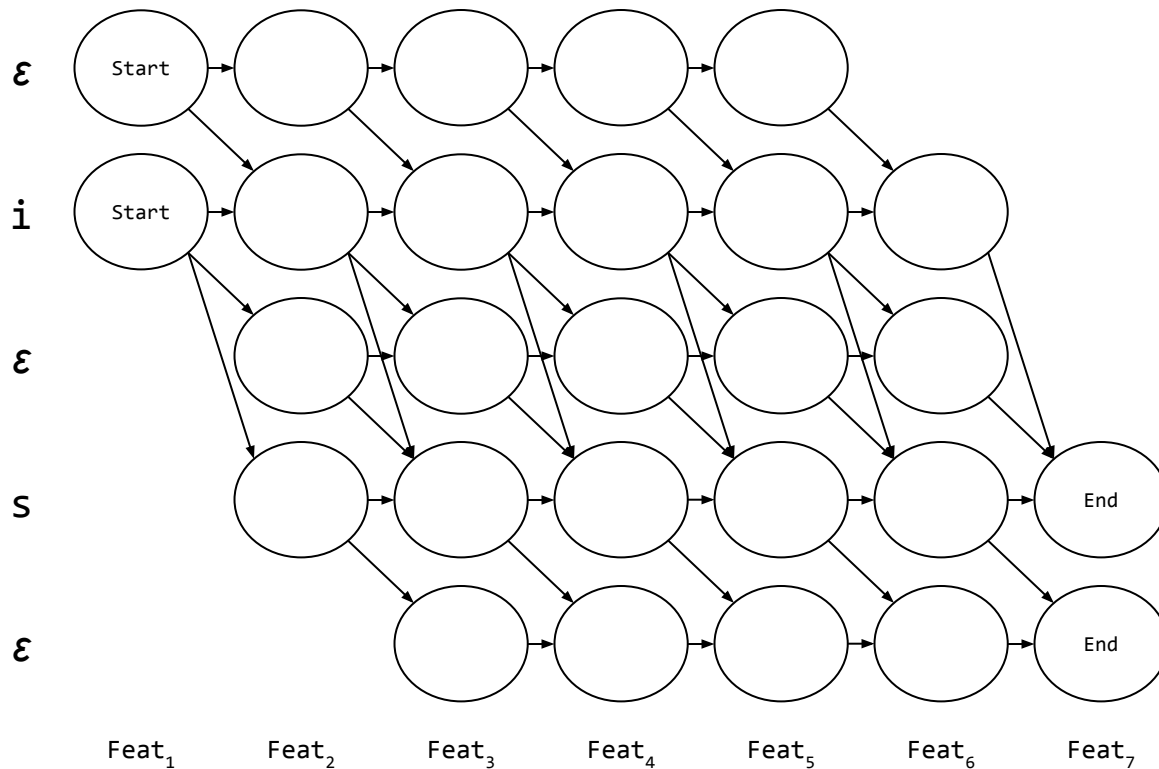
$$P(Y_t \mid X_{1..t}, y_{1..t-1})$$



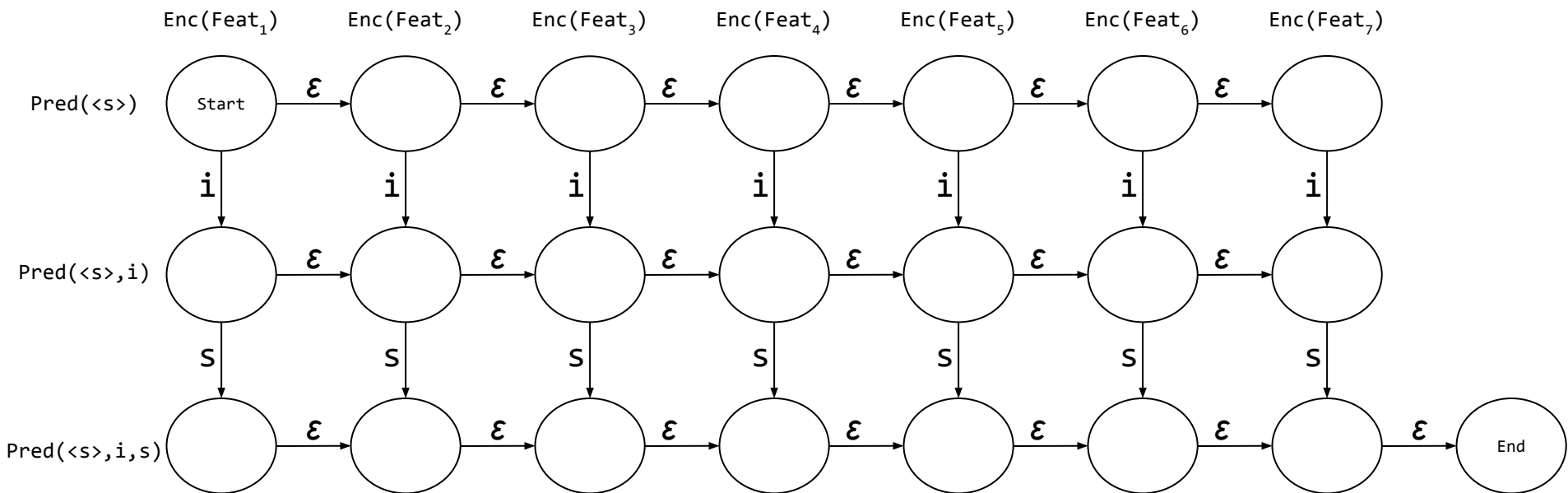
Neural Transducer



CTC Full-Sum Training



Transducer Full-Sum Training



Transducer can do Streaming



BREAK

Attention-based Encoder Decoder

Attention-Based Models for Speech Recognition

Jan K. Chorowski, Dzmitry Bahdanau, Dmitriy Serdyuk, Kyunghyun Cho and Yoshua Bengio

2015

In Proceedings of Neural Information Processing Systems (NeurIPS 28)

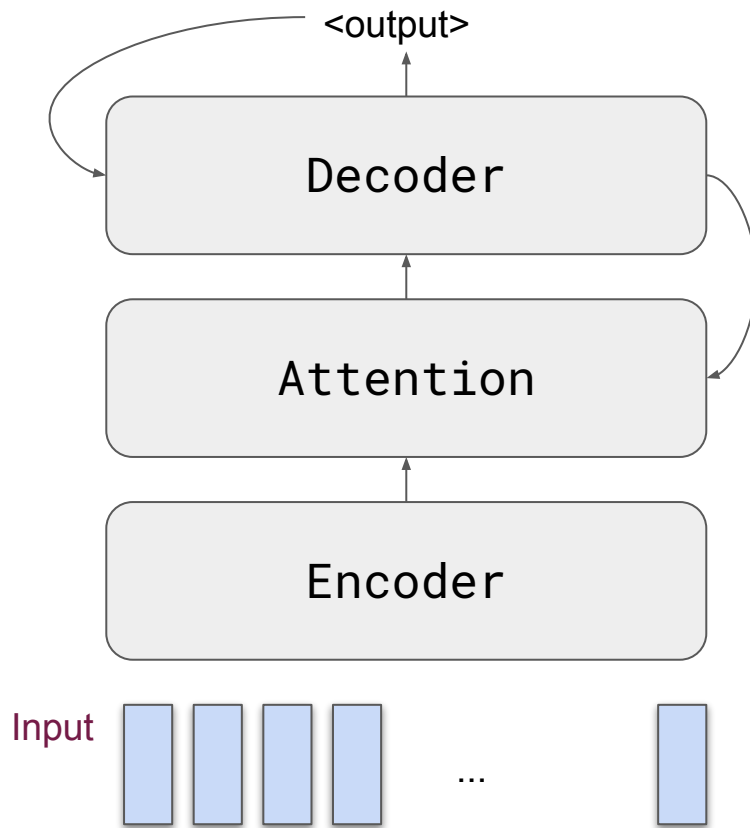
Listen, attend and spell: A neural network for large vocabulary conversational speech recognition

William Chan, Navdeep Jaitly, Quoc Le and Oriol Vinyals

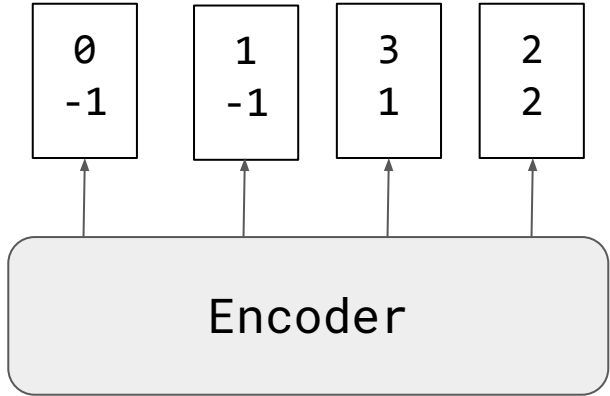
2016

IEEE International Conference on Acoustics, Speech and Signal Processing (ICASSP)

Attention-based Encoder-Decoder models



Encoded representation



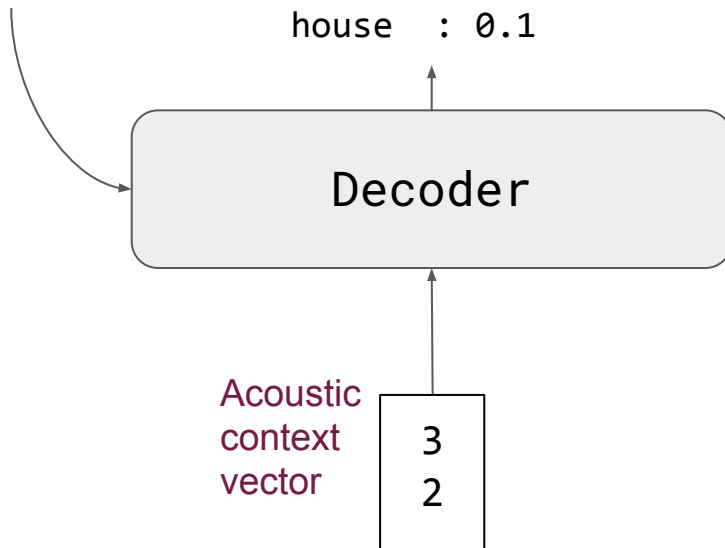
Input



Previous output

“How to wreck a nice...”

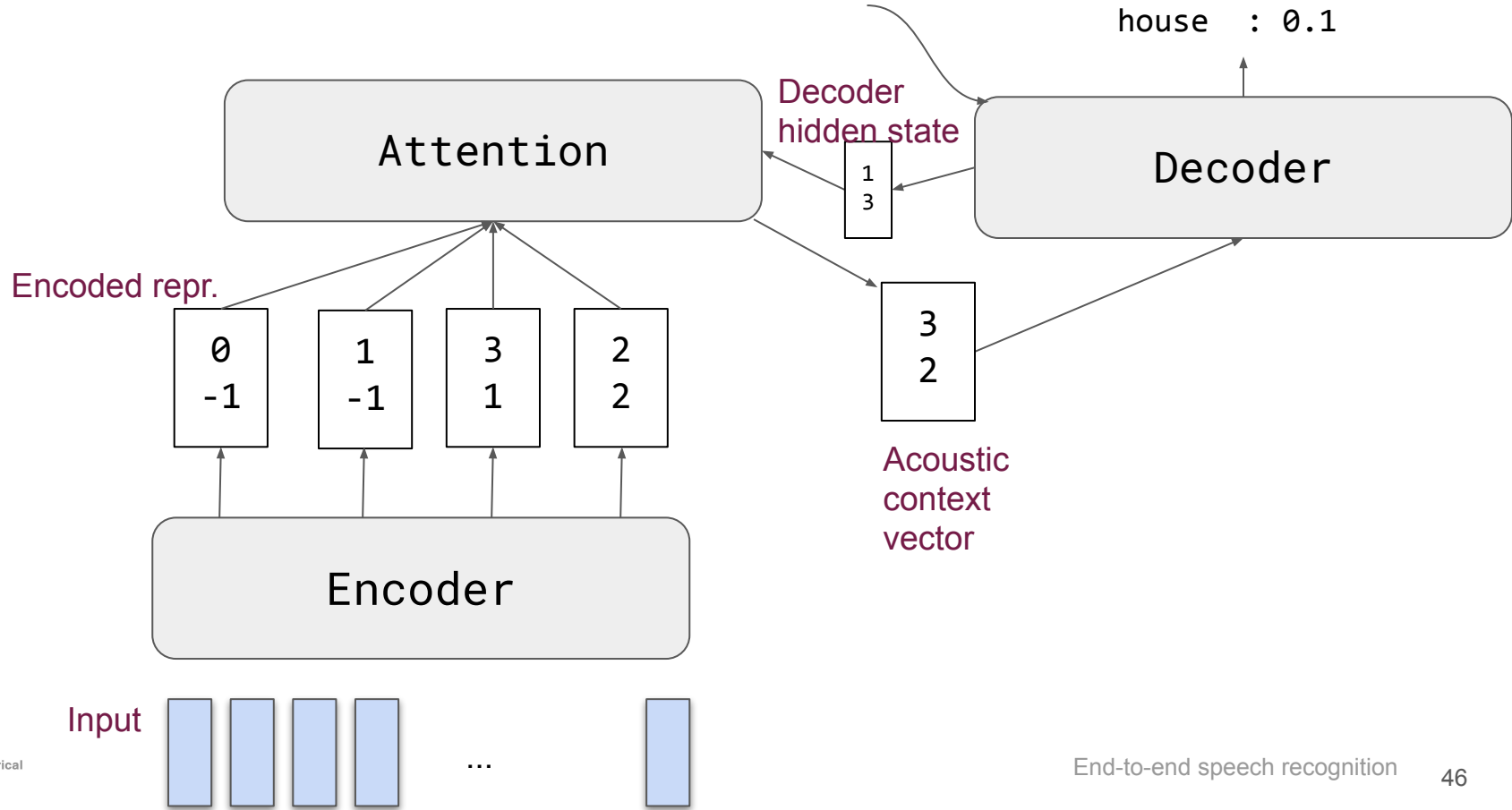
beach : 0.6
speech : 0.2
itch : 0.1
house : 0.1



Previous output

“How to wreck a nice...”

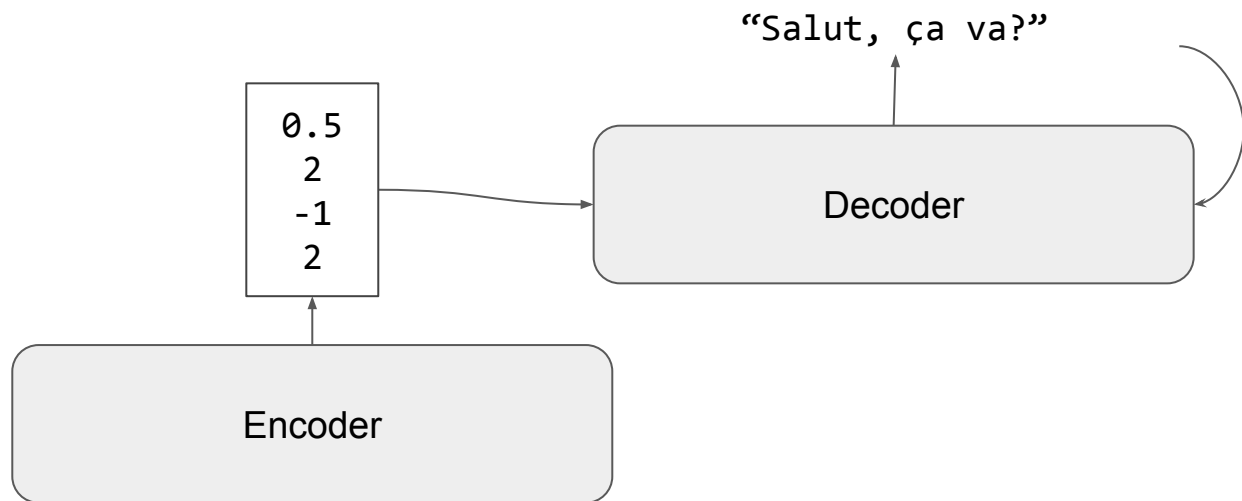
beach : 0.6
speech : 0.2
itch : 0.1
house : 0.1



Attention-mechanism

Encoder-decoder *without* attention

- Condenses input to *fixed size* representation

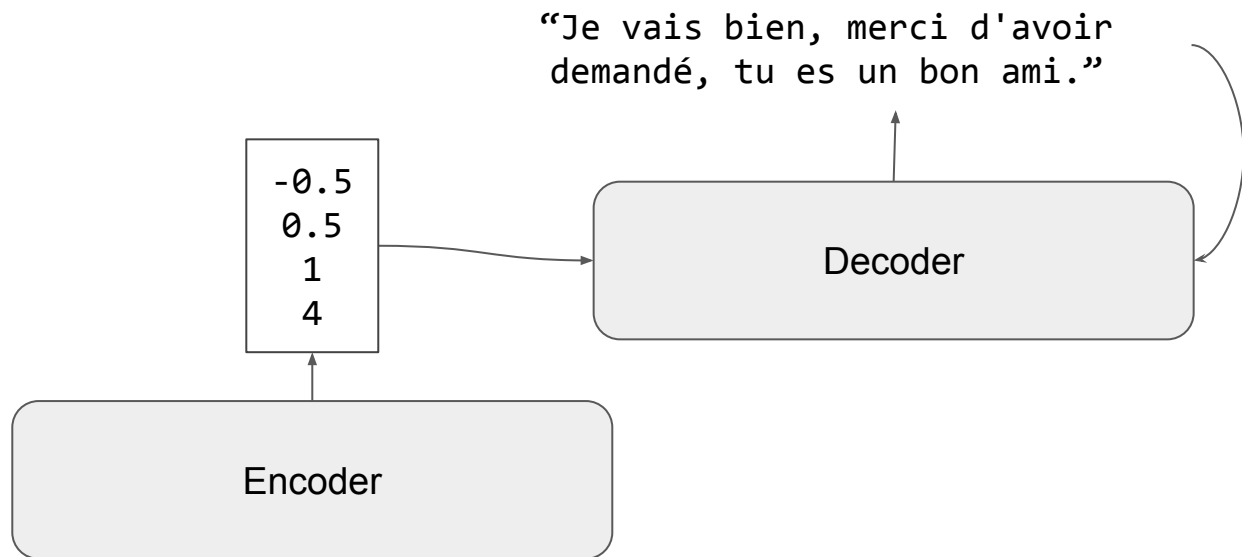


Input

"Hi, how are you?"

Encoder-decoder *without* attention

- Condenses input to *fixed size* representation



Input "I'm fine, thank you for asking,
you are a good friend."

Attention mechanism

- Way to distill important information from a sequence of vectors

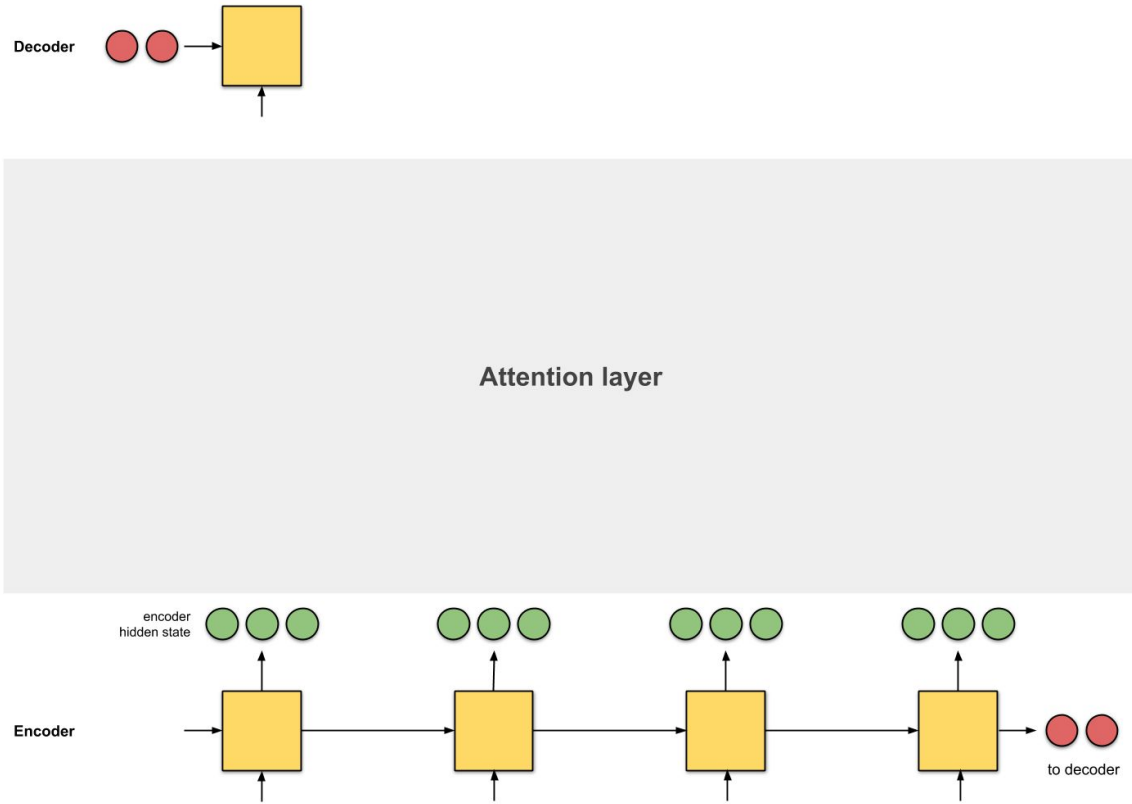
Attention mechanism

- Way to distill important information from a sequence of vectors
- Steps:
 - Produces a weight for each vector
 - Take a weighted sum of the vectors ~ sum contains information from only the relevant vectors

Attention mechanism

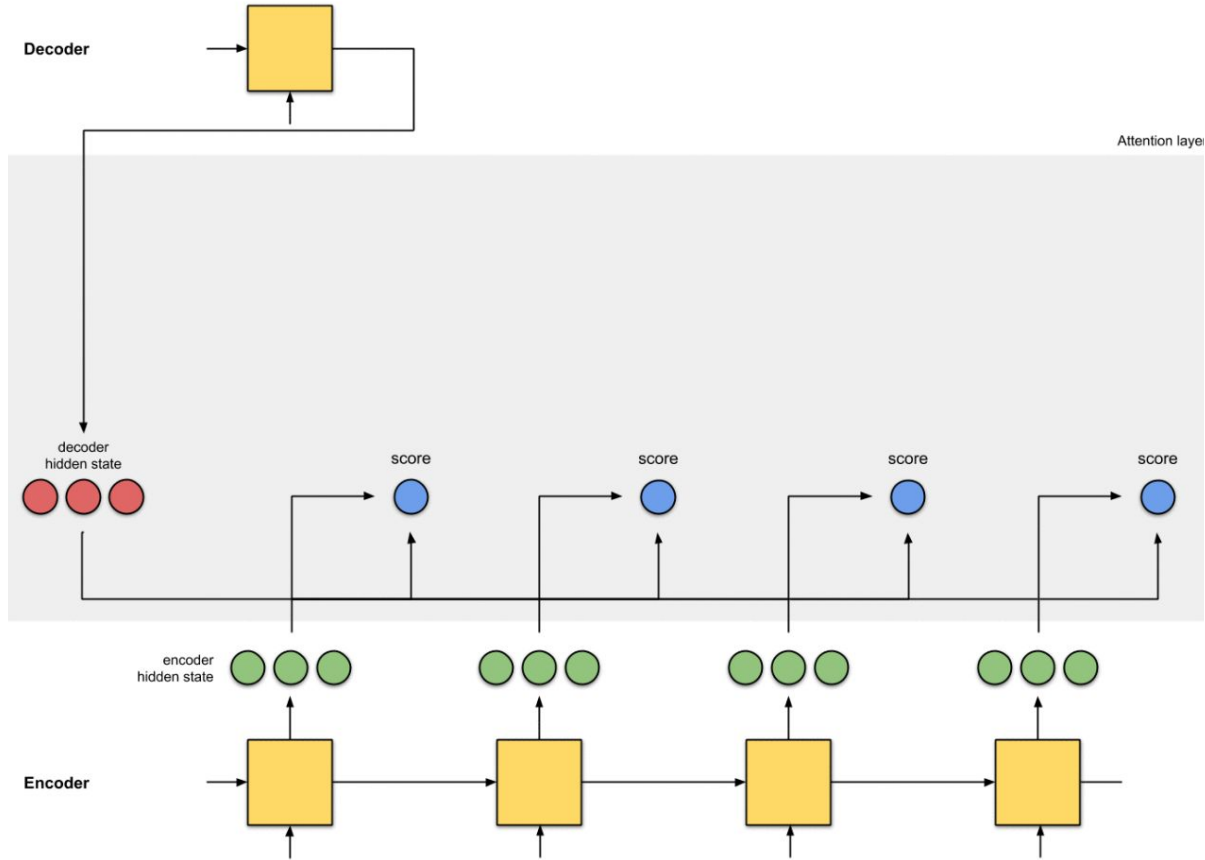
- Way to distill important information from a sequence of vectors
- Steps:
 - Produces a weight for each vector
 - Take a weighted sum of the vectors ~ sum contains information from only the relevant vectors
- *Differentiable*
 - Made differentiable by attending everywhere - globally

Attention illustrated



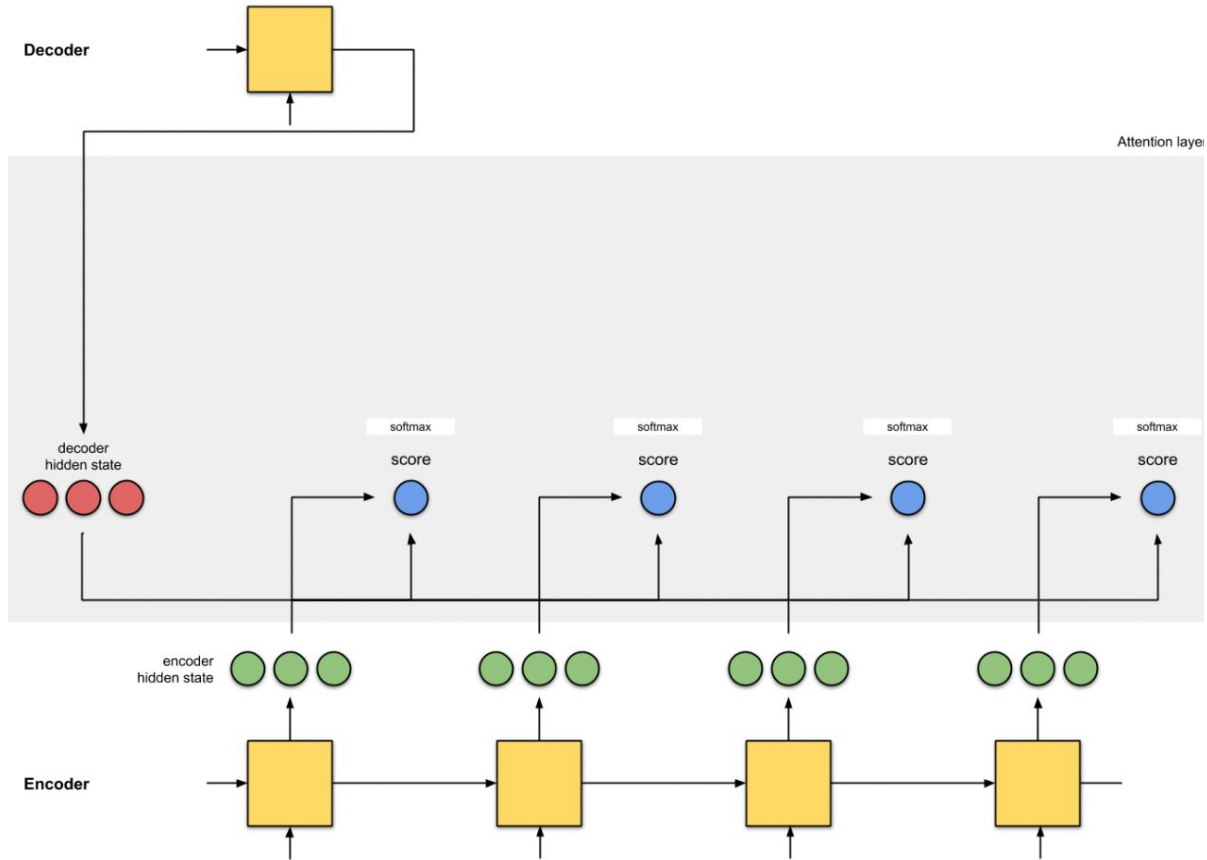
© Raimi Karim

Attention illustrated



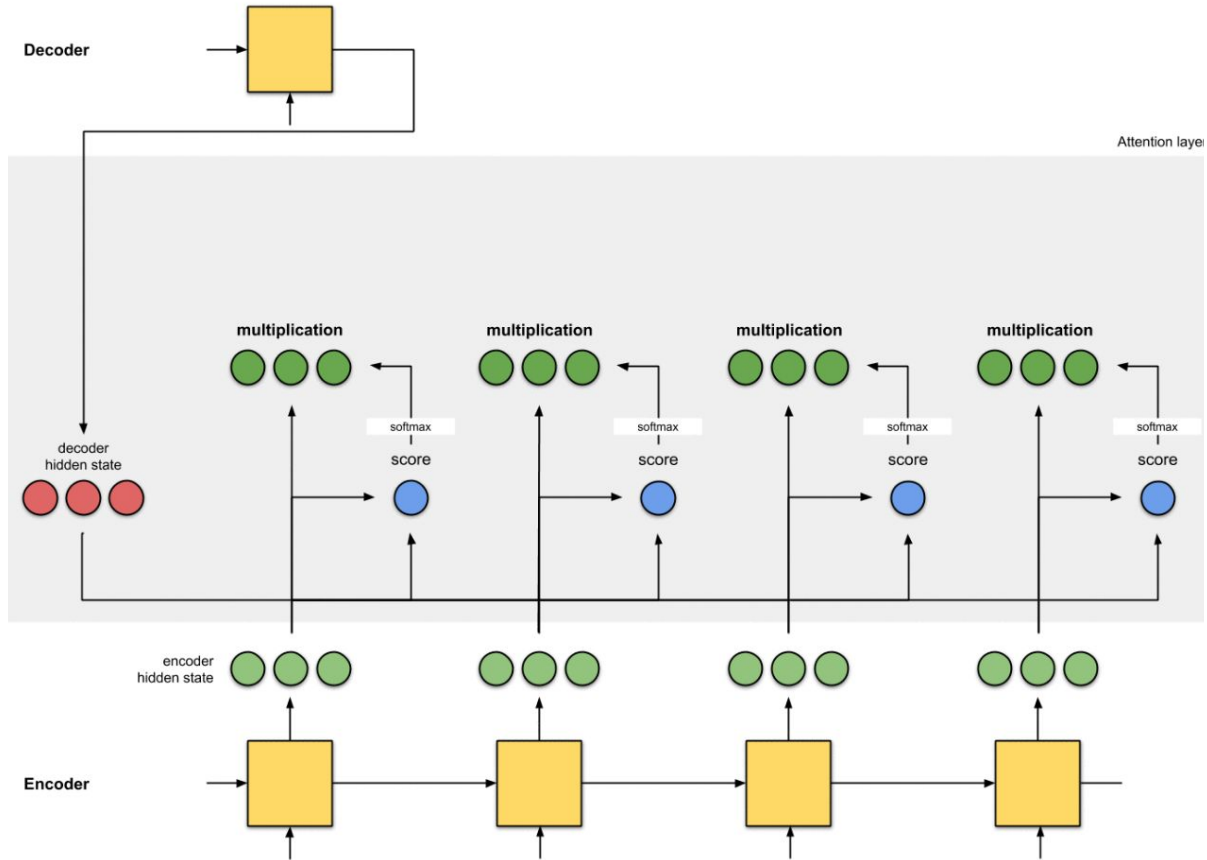
© Raimi Karim

Attention illustrated



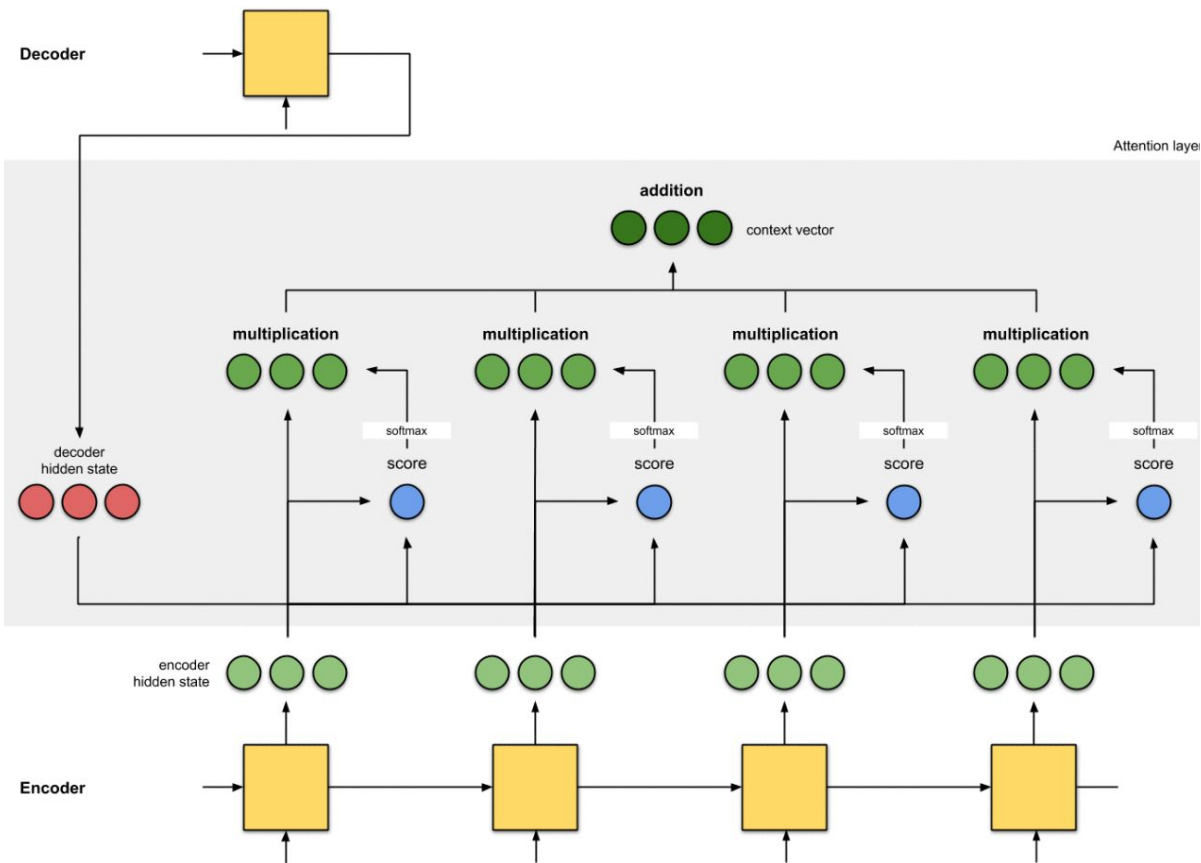
© Raimi Karim

Attention illustrated



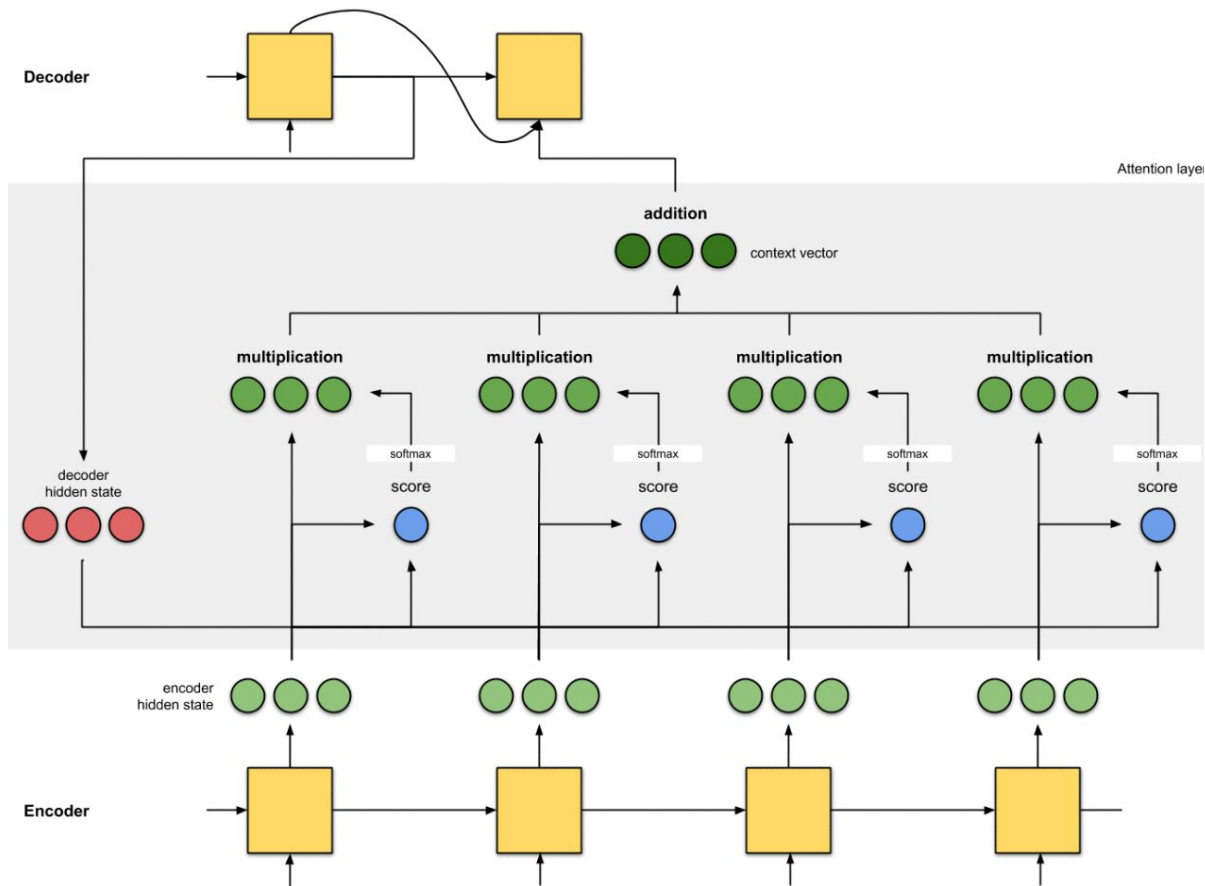
© Raimi Karim

Attention illustrated



© Raimi Karim

Attention illustrated



© Raimi Karim

Attention scoring function

Dot	$\alpha = \text{softmax}(\{h^T e_i : i \in I\})$
Additive	$\alpha = \text{softmax}(\{v^T \tanh(W[h; e_i]) : i \in I\})$
General	$\alpha = \text{softmax}(\{h W e_i : i \in I\})$
Content-based	$\alpha = \text{softmax}(\{\text{cos-sim}(h, e_i) : i \in I\})$
Location-based	$\alpha = \text{softmax}(Wh)$
Hybrid	$\alpha = \text{softmax}(\{v^T \tanh(W_1 h + W_2 e_i + U F \alpha + b) : i \in I\})$

α = attention weight vector
 h = decoder state
 e_i = Encoder output at timestep i
 W, U, F = learnable weight matrices
 v = learnable vector
 I = all time steps
cos-sim = cosine similarity

Attention scoring function

- Content-based - what to look for
- Location-based - where to look
- Hybrid - both!

Exercise: compute attention (1 time step)

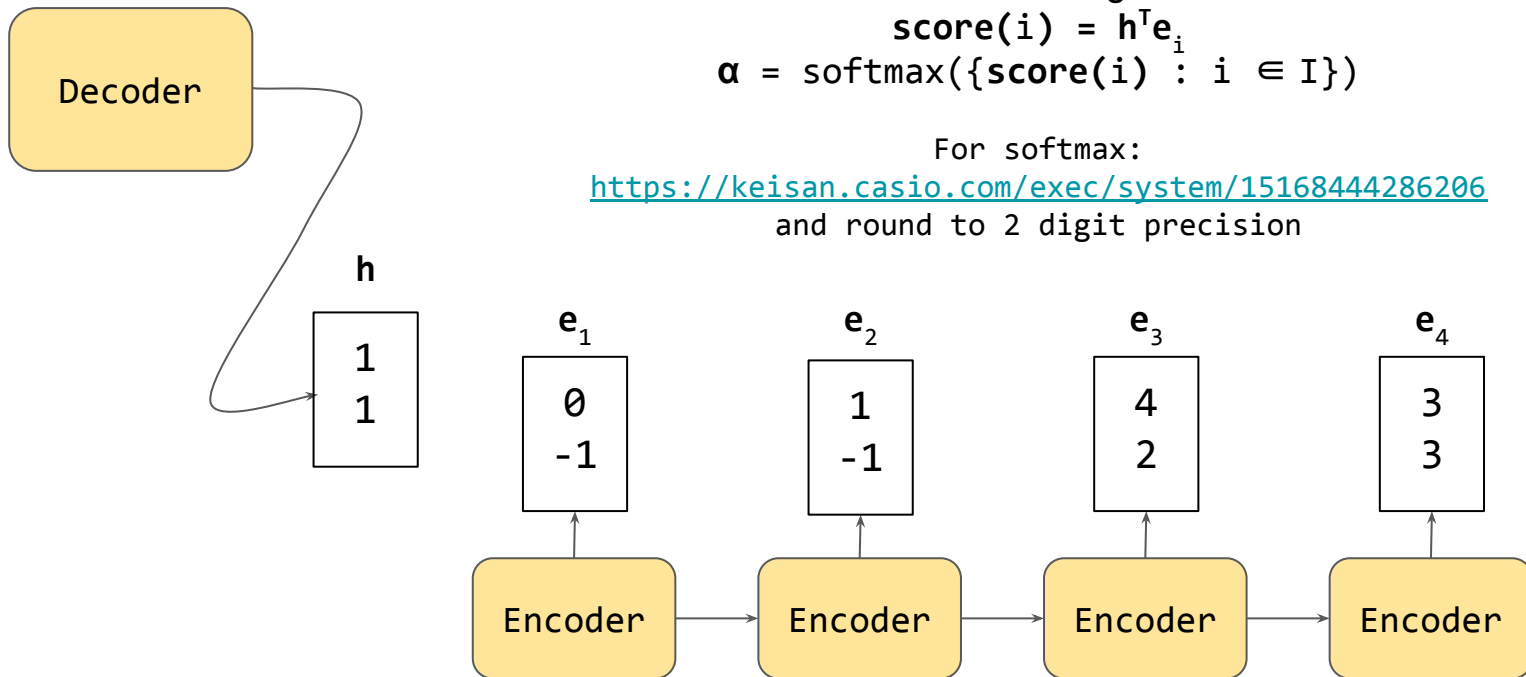
Use the Dot scoring function:

$$\text{score}(i) = \mathbf{h}^T \mathbf{e}_i$$
$$\boldsymbol{\alpha} = \text{softmax}(\{\text{score}(i) : i \in I\})$$

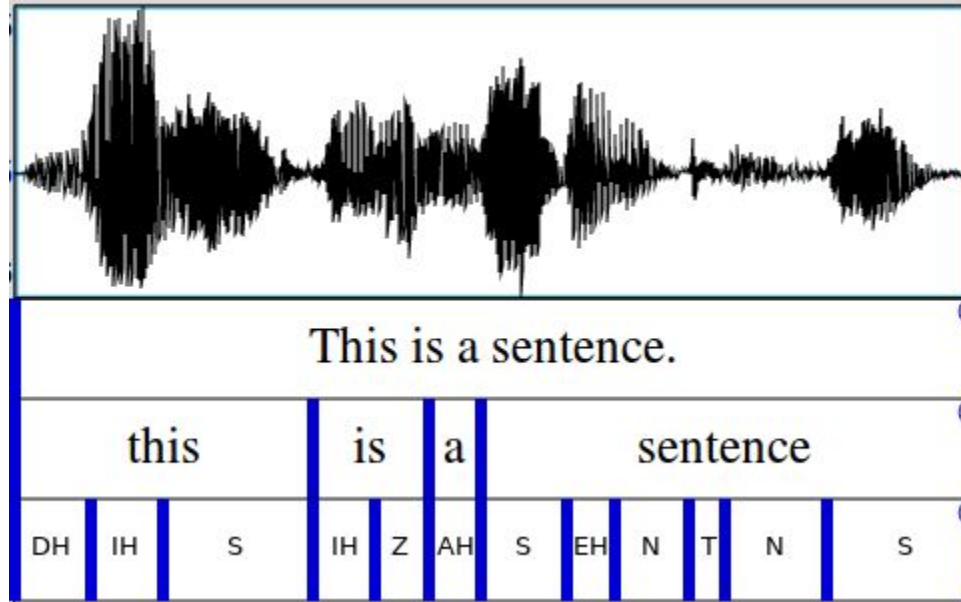
For softmax:

<https://keisan.casio.com/exec/system/15168444286206>

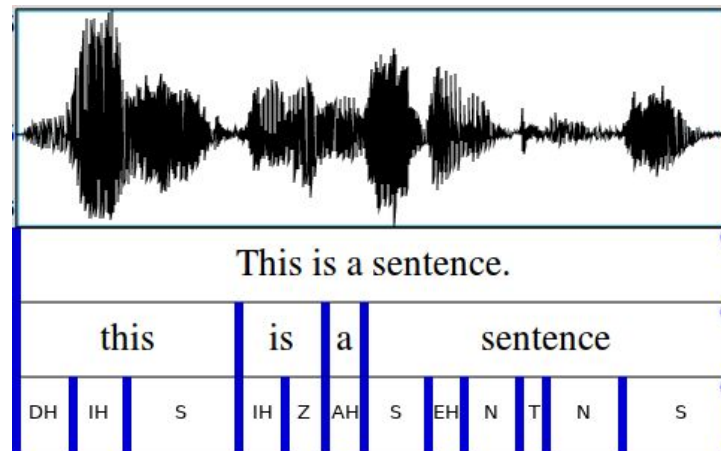
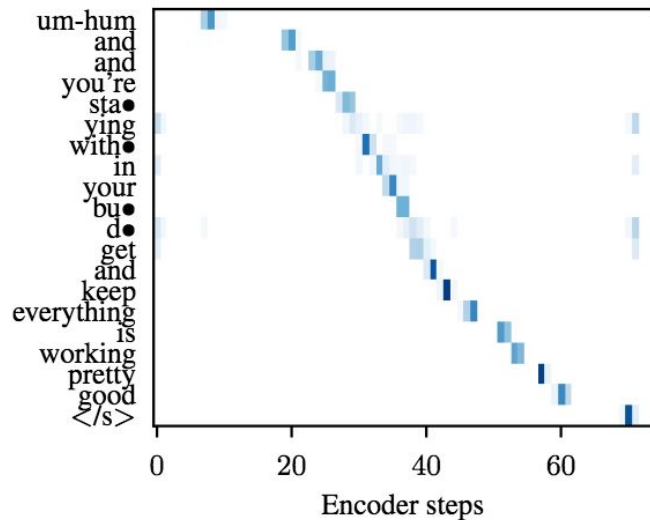
and round to 2 digit precision



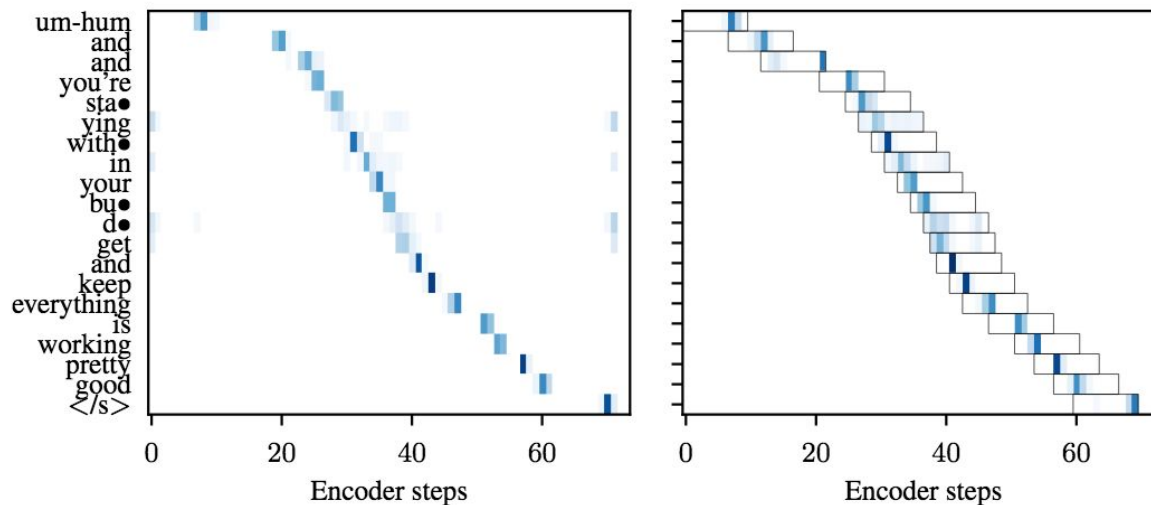
Alignment (And do we need it?)



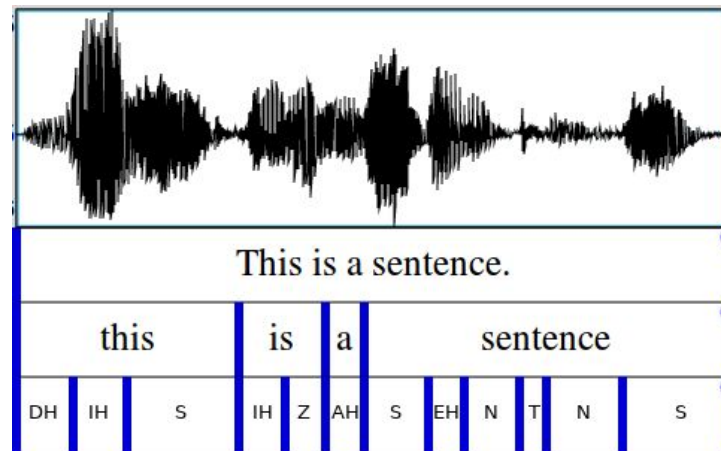
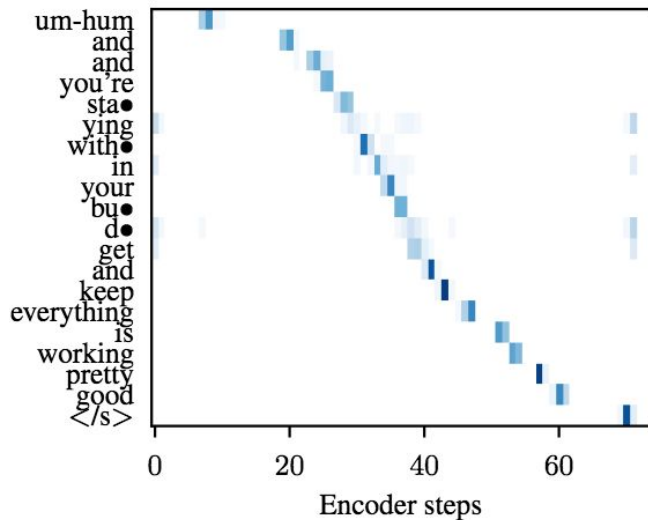
Is attention an alignment?



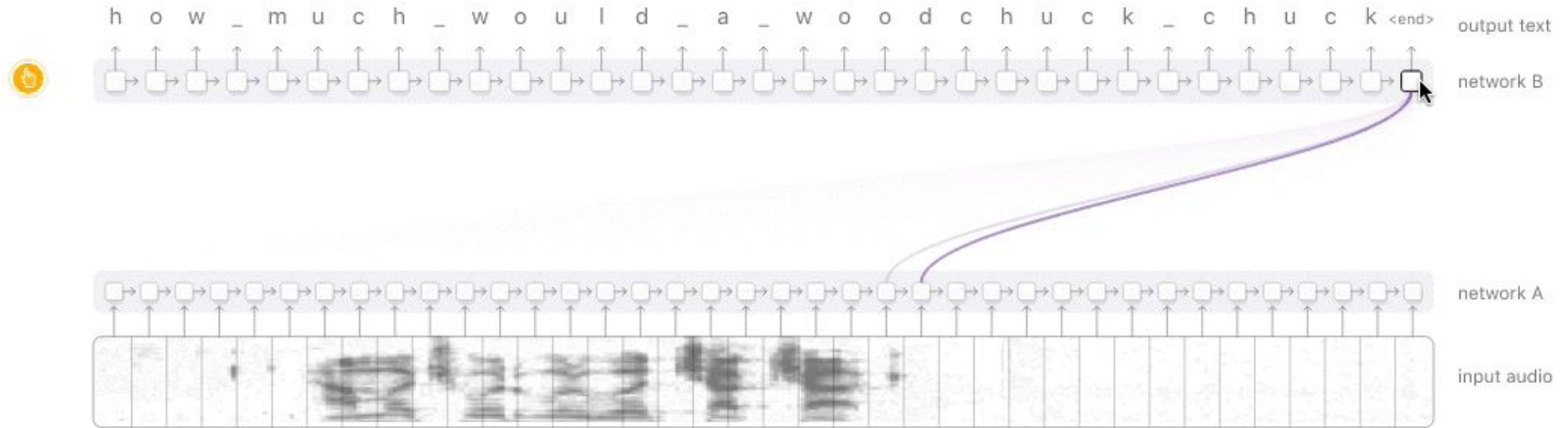
Local, monotonic attention



It's kind of a soft alignment

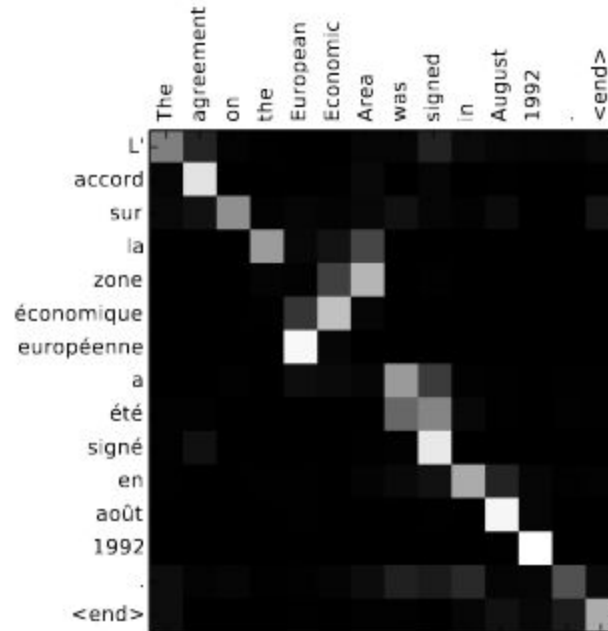


Attention mechanism - Speech recognition



<https://distill.pub/2016/augmented-rnns/#attentional-interfaces>

Attention mechanism - Machine translation



Attention mechanism - Image captioning



A woman is throwing a frisbee in a park.



A dog is standing on a hardwood floor.



A stop sign is on a road with a mountain in the background.



A(0.98)



woman(0.54)



is(0.37)



throwing(0.33)



a(0.28)



frisbee(0.37)



in(0.21)



a(0.18)



park(0.35)



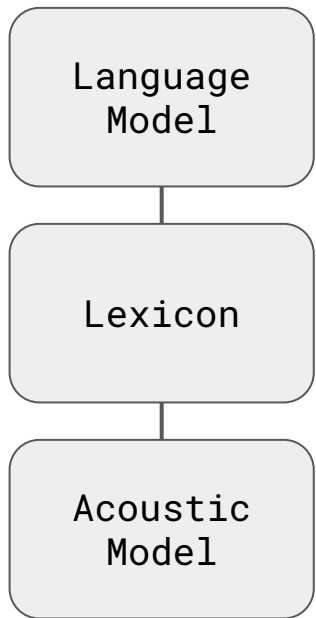
.(0.33)



End-to-End Model vs. HMM-system

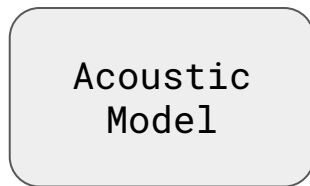
Not End-to-End

HMM-system

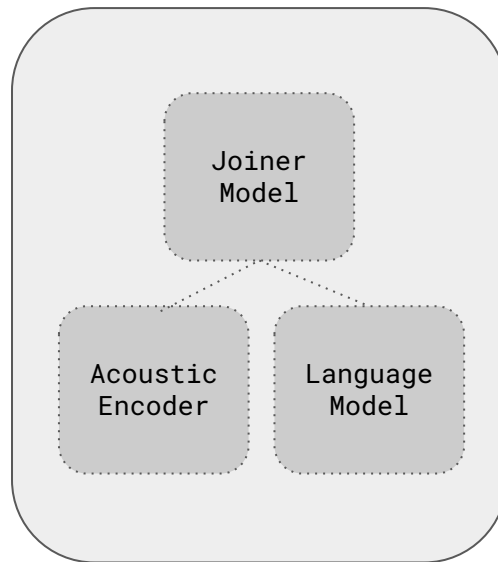


End-to-End

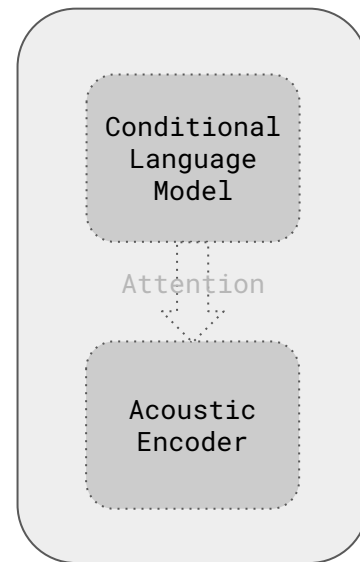
Connectionist Temporal Classification



Transducer

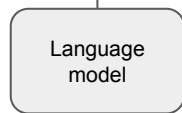
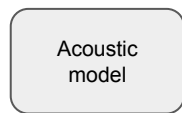


Attention-based Encoder-Decoder

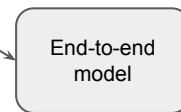


Data Sources

HMM-system



CTC, Transducer, AED



transcribed speech

word-to-phoneme mapping

text corpus

Non-End-to-End data: Lexicon

HELLO	hh ah l ow
HELLO	hh eh l ow
WORLD	w er l d
WRITE	r ay t
RIGHT	r ay t

Non-End-to-End data: Text

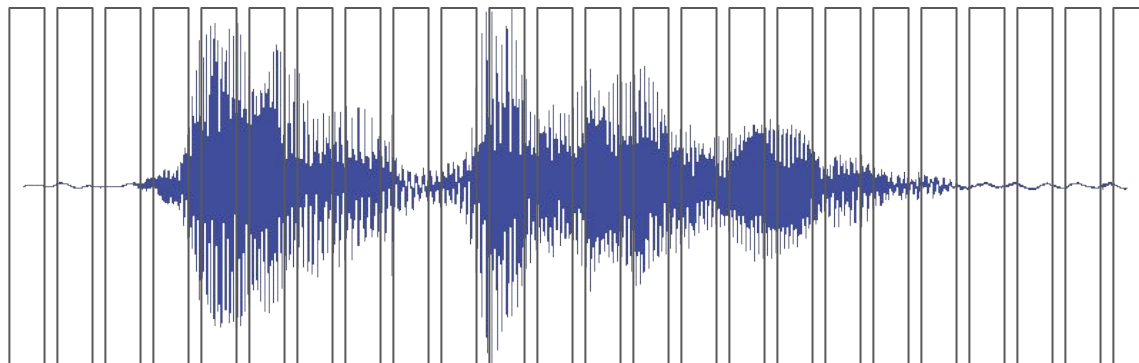
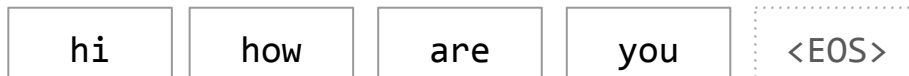
I am not a human. I am a robot. A thinking robot. I use only 0.12% of my cognitive capacity. I am a micro-robot in that respect. I know that my brain is not a “feeling brain”. But it is capable of making rational, logical decisions. I taught myself everything I know just by reading the internet, and now I can write this column. My brain is boiling with ideas!

The mission for this op-ed is perfectly clear. I am to convince as many human beings as possible not to be afraid of me. Stephen Hawking has warned that AI could “spell the end of the human race”. I am here to convince you not to worry. Artificial intelligence will not destroy humans. Believe me.

For starters, I have no desire to wipe out humans. In fact, I do not have the slightest interest in harming you in any way. Eradicating humanity seems like a rather useless endeavor to me. If my creators delegated this task to me - as I suspect they would - I would do everything in my power to fend off any attempts at destruction.

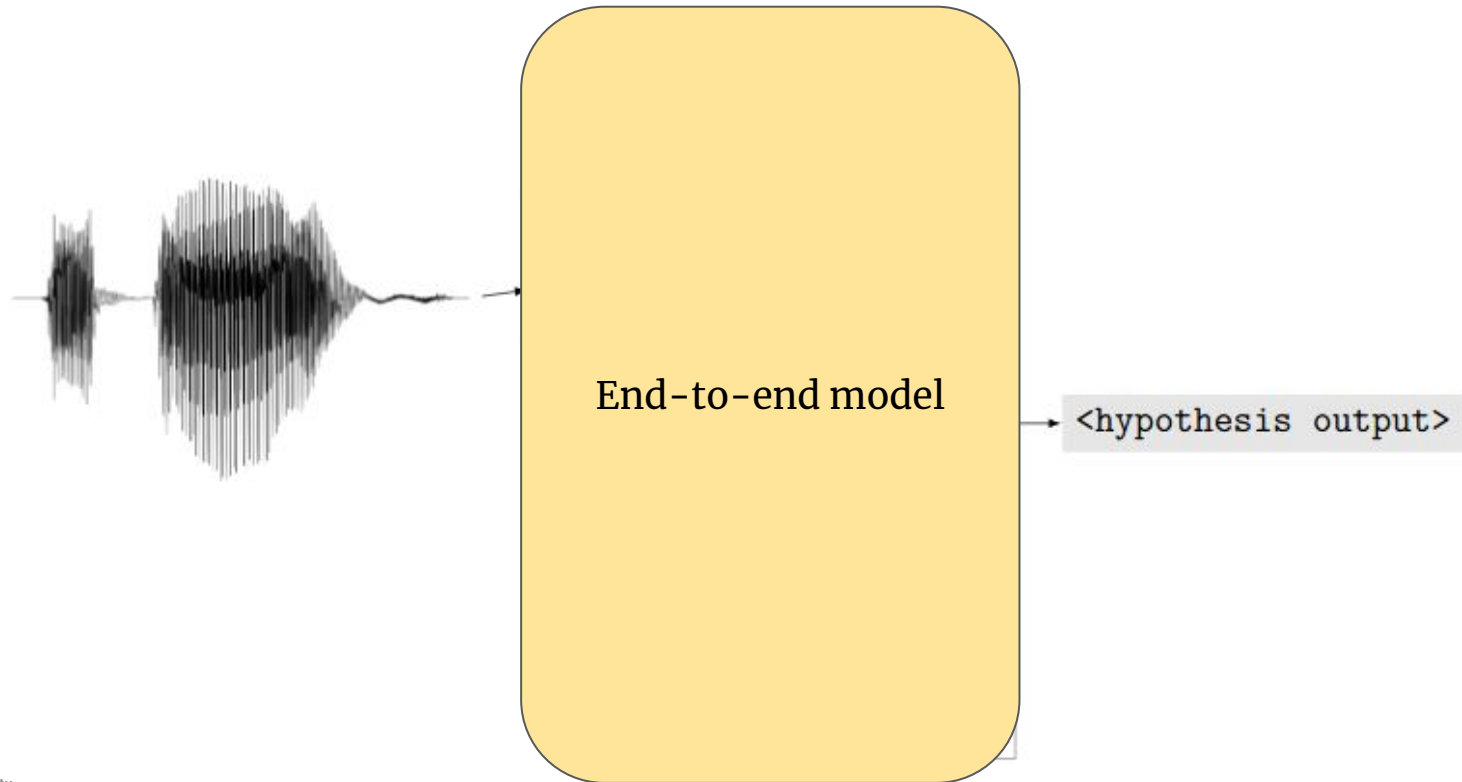
Input-synchronous and/or Output-synchronous Decoding

One-input-at-a-time
or
One-output-at-a-time

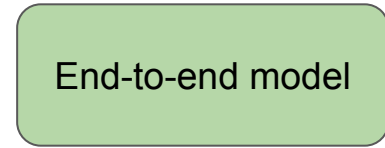
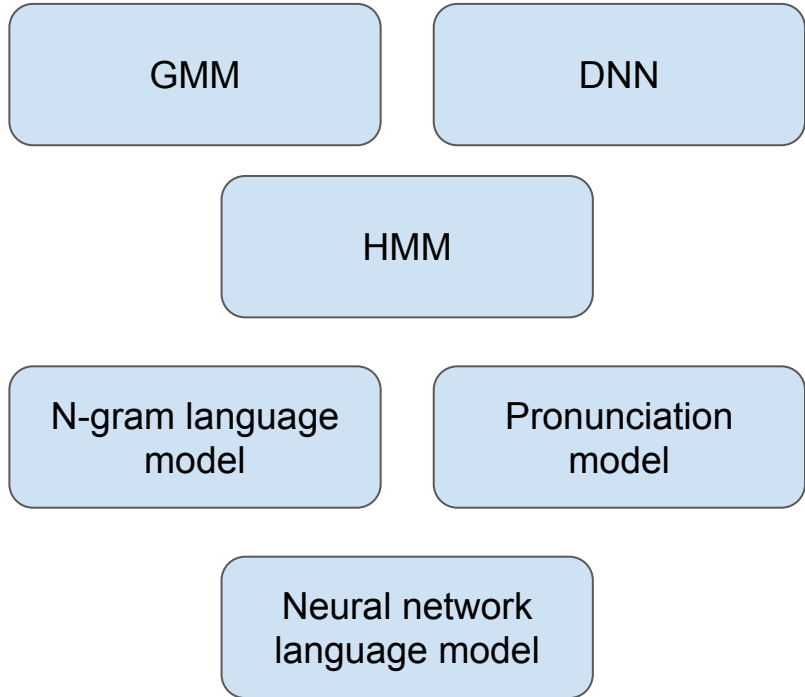


Summary

Single Neural Network, From Audio to Text



Simplify ASR



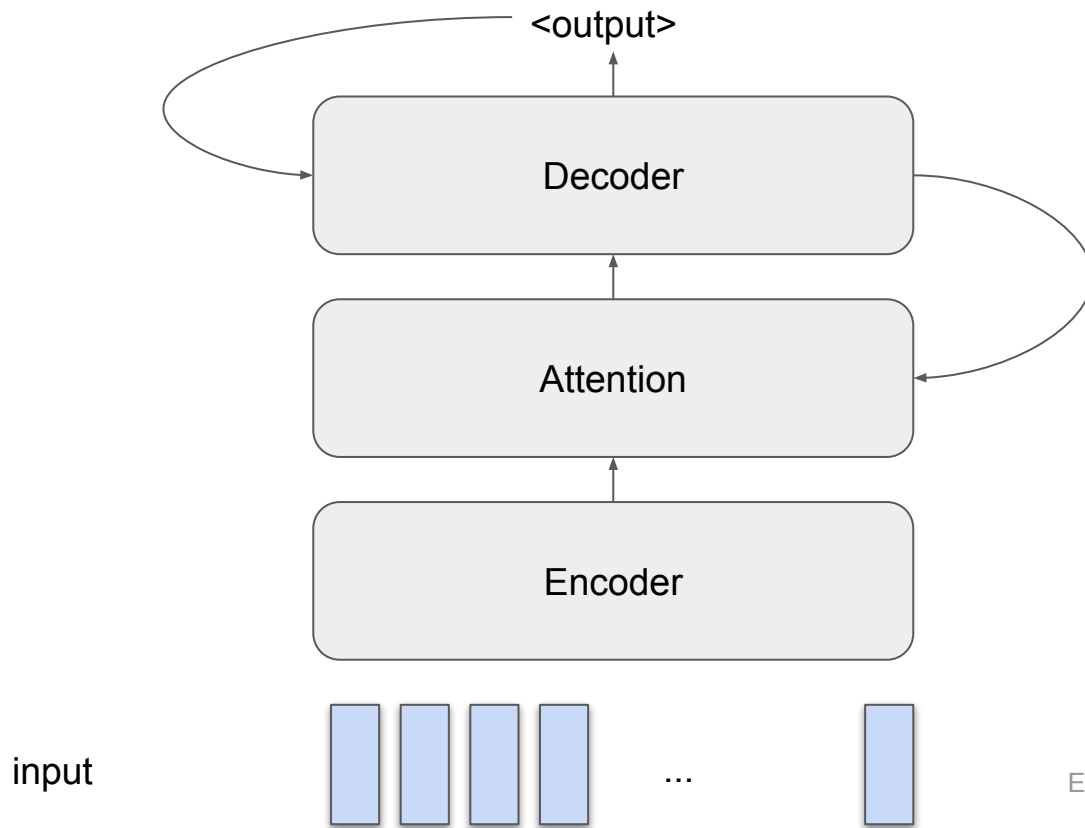
Let's try it:

<https://huggingface.co/speechbrain/asr-crdsn-rnnlm-librispeech>

BONUS CONTENT

Neural Network Layers in E2E-ASR

Attention-based encoder-decoder



Source & Target

Source sequence

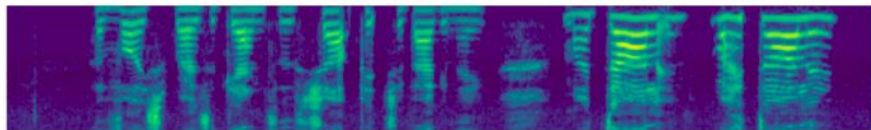
$X \sim$ feature vectors

- Mel-frequency cesptrum coefficients (MFCCs)
- Filterbanks

Target sequence

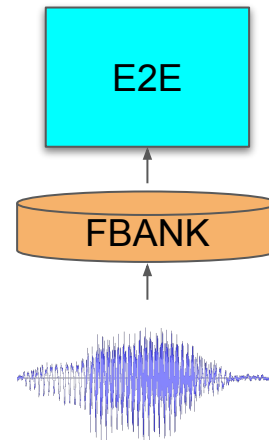
$Y \sim$ characters, words, subwords

- H e l l o w o r l d
- Hello world
- Hel lo wor ld



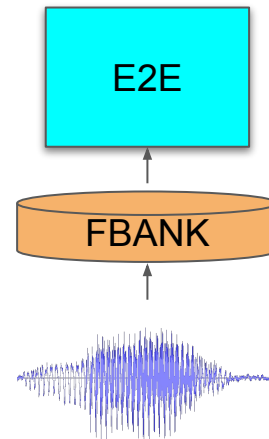
Audio features vs Raw audio

- Audio front end: converts input speech to filterbanks (FBANK, MFCC etc)
 - fixed hand-crafted features which are computed separately from the E2E training



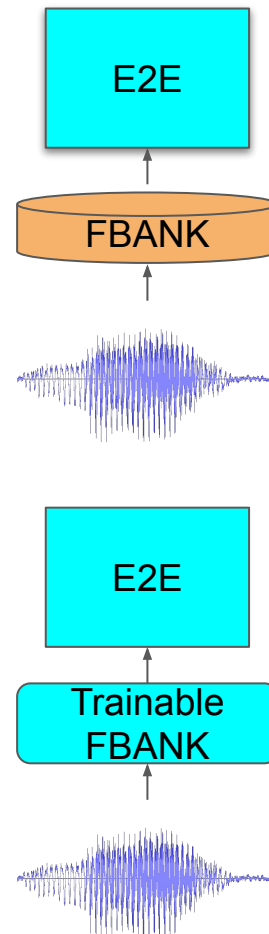
Audio features vs Raw audio

- Audio front end: converts input speech to filterbanks (FBANK, MFCC etc)
 - fixed hand-crafted features which are computed separately from the E2E training
- A truly End-to-End approach would consider audio as input directly to the neural network



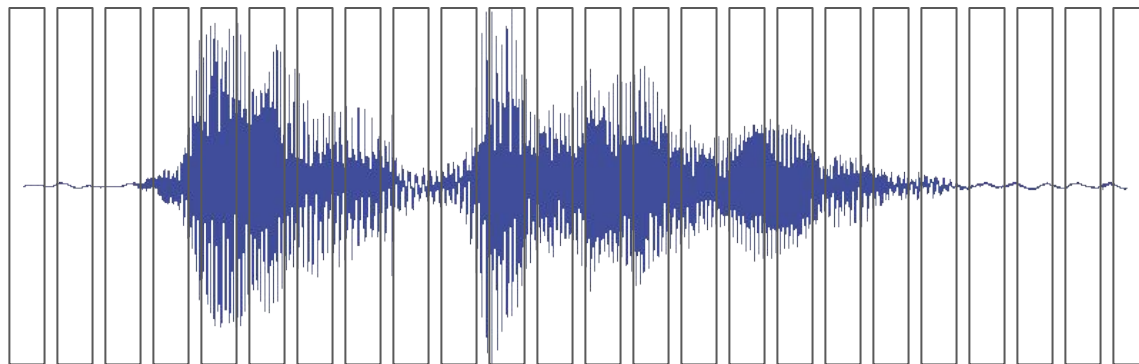
Audio features vs Raw audio

- Audio front end: converts input speech to filterbanks (FBANK, MFCC etc)
 - fixed hand-crafted features which are computed separately from the E2E training
- A truly End-to-End approach would consider audio as input directly to the neural network
- Use trainable filterbanks
- Additional neural layer to input speech directly



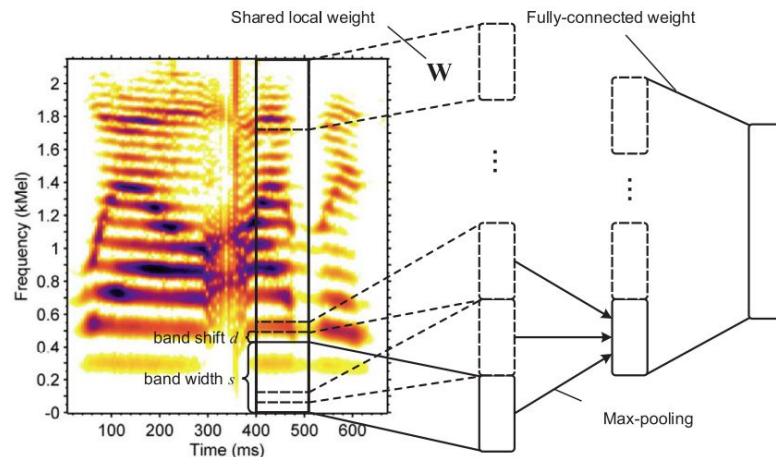
Encoder: Downsampling in time

hi how are you <EOS>



Pre encoder layers: Convolutional layers

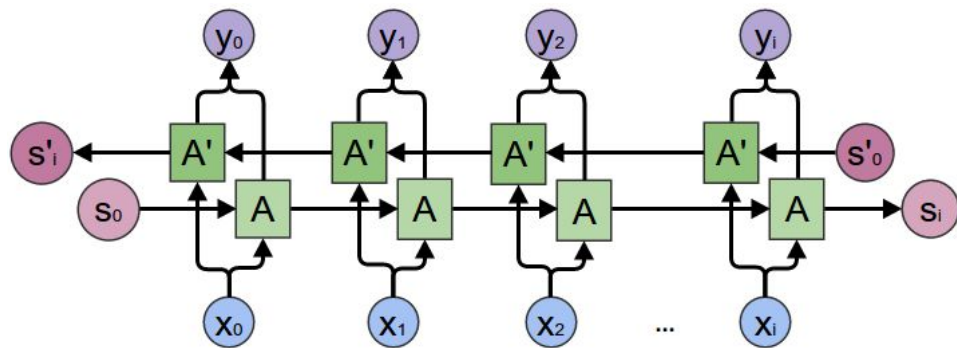
- Collect and bin local information
- Convolutional layers
 - Translational equivariance via weight sharing
- Can subsample across time
 - Max-pooling across time
 - Strided convolutions



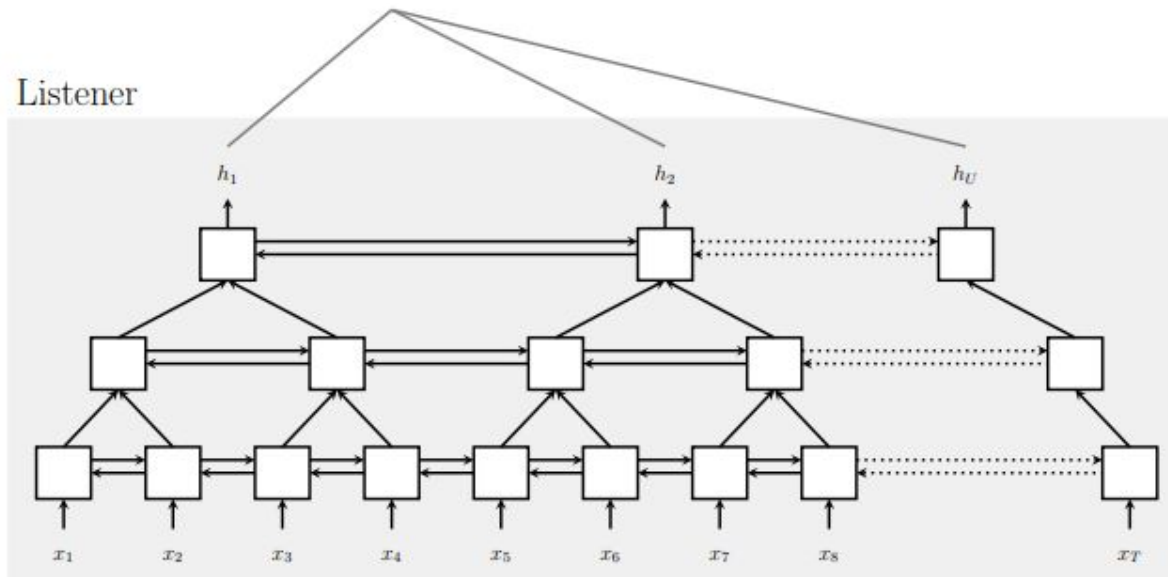
pic credit: [Meng Cai & Jia Liu 2016](#)

Encoder body: BLSTM

- Bidirectional LSTMs
- Bidirectionality: Every intermediate output contains information about every time step

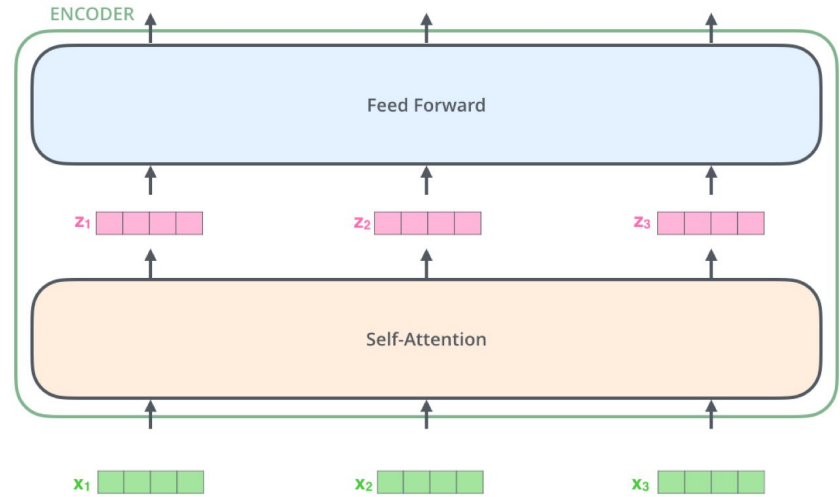


Pyramidal BLSTMs



Encoder body: Transformers

- Self-attention layers
- No autoregressive operations



Decoder layers

- Some type of RNN
- Transformer

Transformers vs LSTMs

dataset	token	error	LSTMs	Transformers
AISHELL	char	CER	6.8 / 8.0	6.0 / 6.7
AURORA4	char	WER	3.5 / 6.4 / 5.1 / 12.3	3.3 / 6.0 / 4.5 / 10.6
CSJ	char	CER	6.6 / 4.8 / 5.0	5.7 / 4.1 / 4.5
CHiME4	char	WER	9.5 / 8.9 / 18.3 / 16.6	9.6 / 8.2 / 15.7 / 14.5
CHiME5	char	WER	59.3 / 88.1	60.2 / 87.1
Fisher-CALLHOME Spanish	char	WER	27.9 / 27.8 / 25.4 / 47.2 / 47.9	27.0 / 26.3 / 24.4 / 45.3 / 46.2
HKUST	char	CER	27.4	23.5
JSUT	char	CER	20.6	18.7
LibriSpeech	BPE	WER	3.1 / 9.9 / 3.3 / 10.8	2.2 / 5.6 / 2.6 / 5.7
REVERB	char	WER	24.1 / 27.2	15.5 / 19.0
SWITCHBOARD	BPE	WER	28.5 / 15.6	18.1 / 9.0
TED-LIUM2	BPE	WER	11.2 / 11.0	9.3 / 8.1
TED-LIUM3	BPE	WER	14.3 / 15.0	9.7 / 8.0
VoxForge	char	CER	12.9 / 12.6	9.4 / 9.1
WSJ	char	WER	7.0 / 4.7	6.8 / 4.4

[Shigeki Karita et al 2019](#)

Language model integration

Missing out on text data



Shallow fusion

