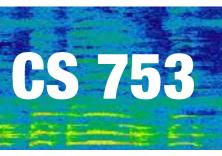
# **End-to-end Neural Architectures For ASR**



Instructor: Preethi Jyothi

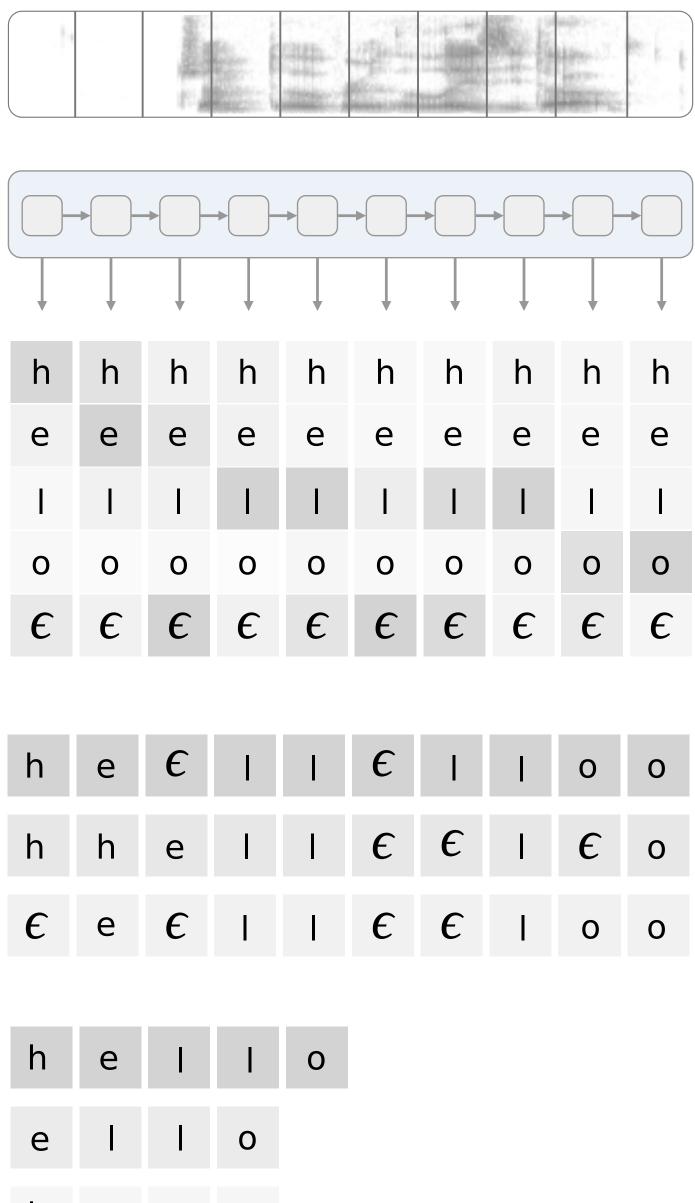
Lecture 15



## **Connectionist Temporal Classification (CTC): Recap**

•

ullet

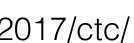


h e 0 CTC objective function is the probability of an output label sequence y given an utterance x (by summing over all possible alignments for y provided by  $B^{-1}(y)$ :

$$CTC(x, y) = \Pr(y \mid x) = \sum_{a \in B^{-1}(y)} \Pr(a \mid x)$$
$$= \sum_{a \in B^{-1}(y)} \prod_{t=1}^{T} \Pr(a_t \mid x)$$

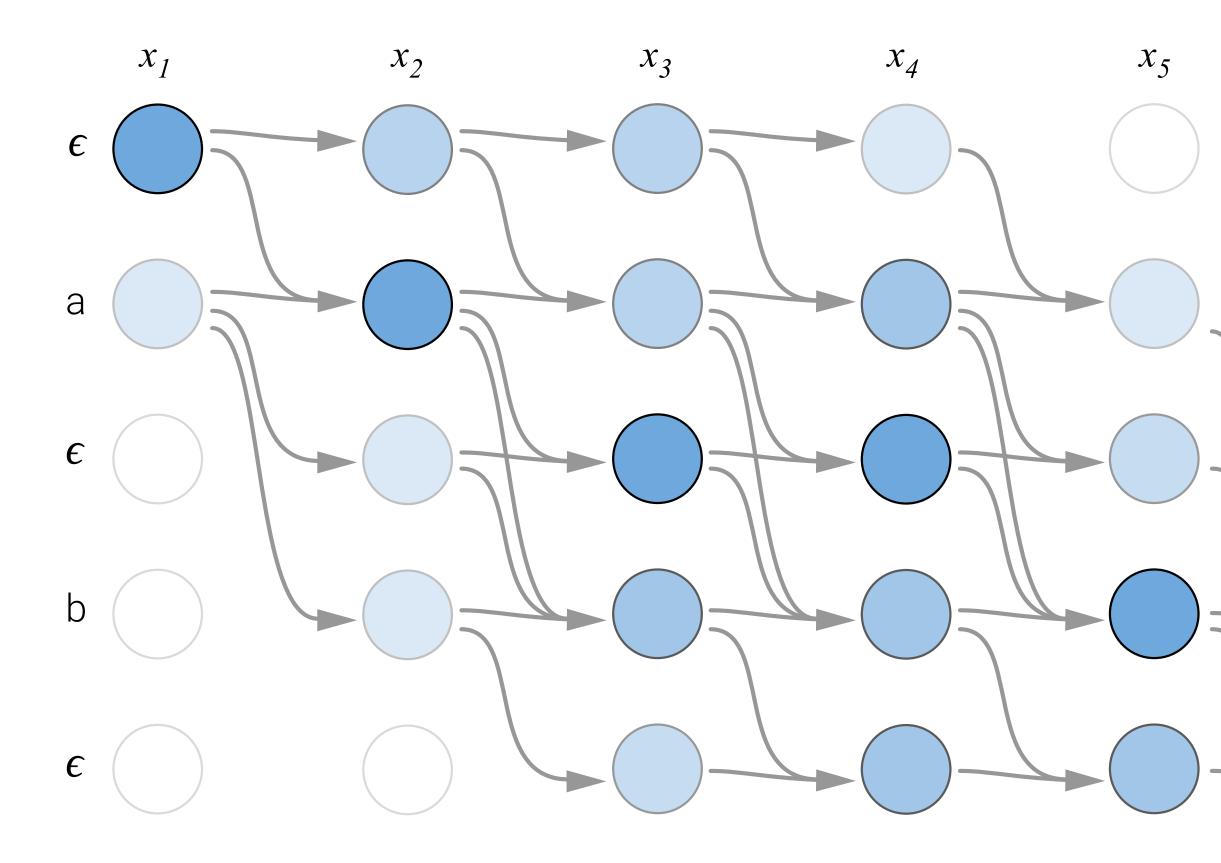
Efficient forward+backward algorithm to compute this loss function and its gradients

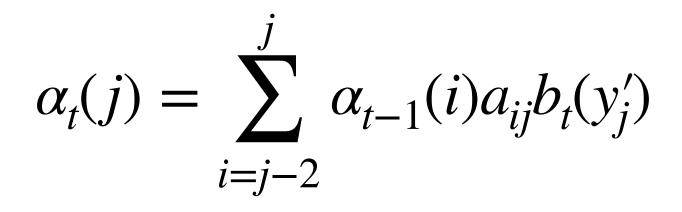




## Illustration: Forward Algorithm to compute $\alpha_t(j)$

 $x_6$ 



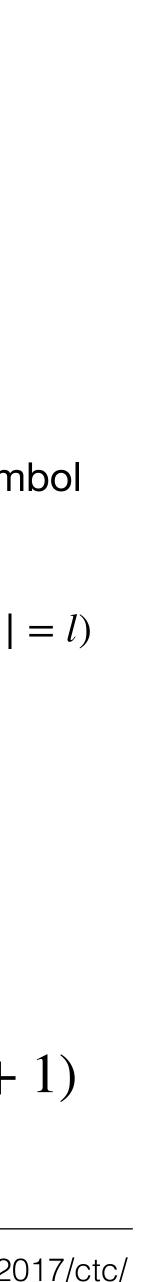


#### where

 $b_t(y'_j)$  is the probability given by NN to the symbol  $y'_j$  for t = 1...T, when |x| = T  $y'_j = \begin{cases} y_{j/2} \text{ if } j \text{ is even} \\ e \text{ otherwise} \end{cases} (j = 1...2l + 1 \text{ when } |y| = l)$ (1 if i = i or i = i - 1)

$$a_{ij} = \begin{cases} 1 \text{ if } i = j \text{ or } i = j - 1 \\ 1 \text{ if } i = j - 2 \text{ and } y'_j \neq y'_{j-2} \\ 0 \text{ otherwise} \end{cases}$$

 $CTC(x, y) = \sum_{a \in B^{-1}(y)} \Pr(a \,|\, x) = \alpha_T(2l) + \alpha_T(2l+1)$ 



## **CTC vs. LAS**

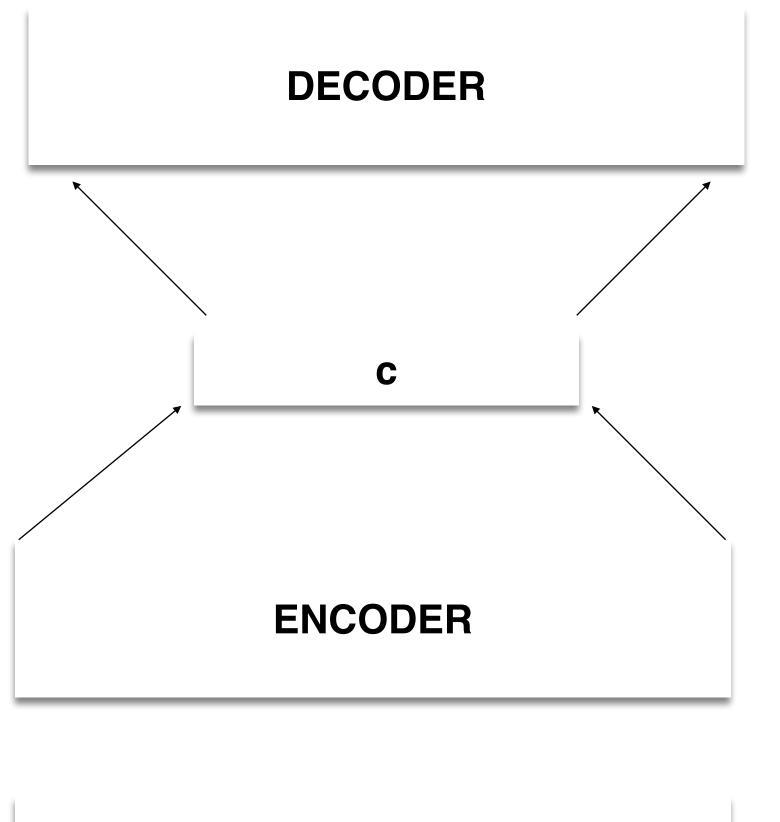
- Works well for end-to-end ASR systems
- CTC makes an assumption that the network outputs at different time steps are conditionally independent given the inputs
- The Listen, Attend and Spell [LAS] network makes no independence assumptions about the probability distribution of the output sequences given the input

 $P(\mathbf{y}|\mathbf{x}) =$ 

Based on the sequence-to-sequence with attention framework

$$\prod_i P(y_i | \mathbf{x}, y_{< i})$$

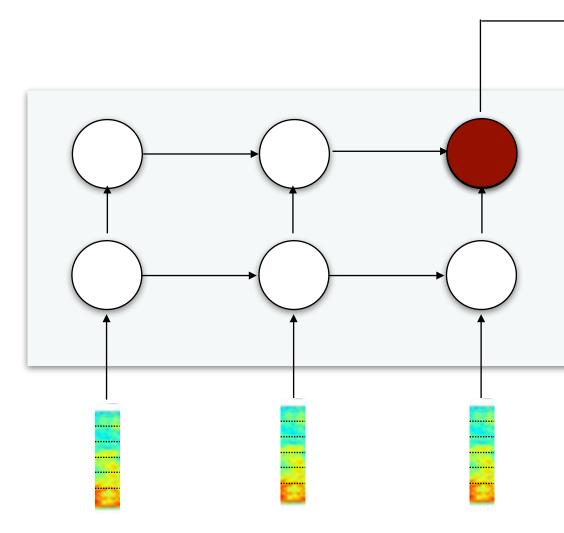
#### Sequence to sequence models Encoder-decoder architecture

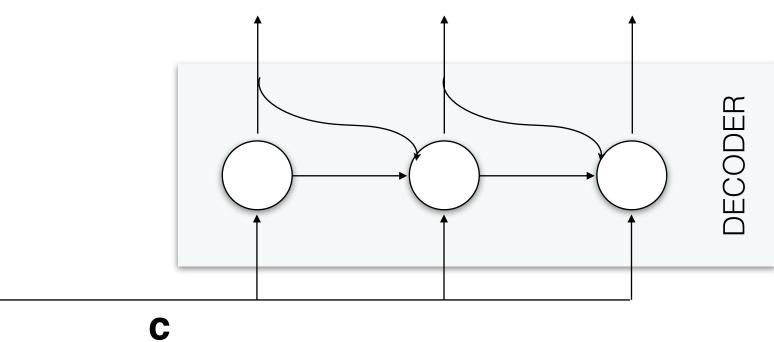


Data

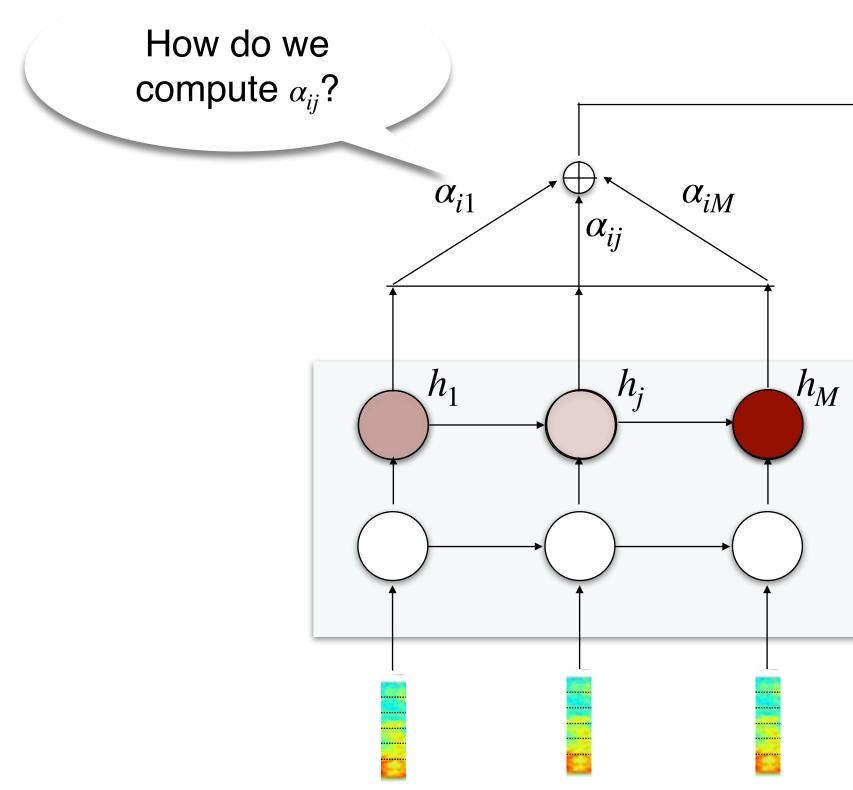
#### Sequence to sequence models Encoder-decoder architecture

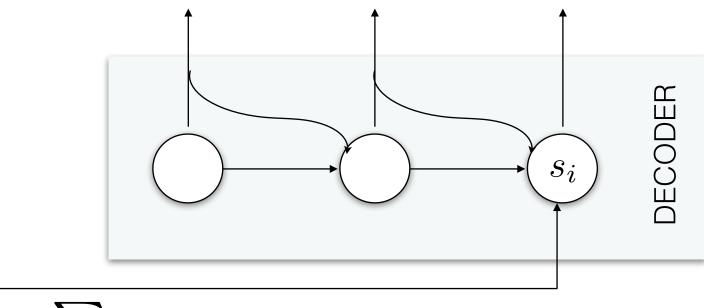
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#### **Sequence to sequence models** Encoder-decoder with attention

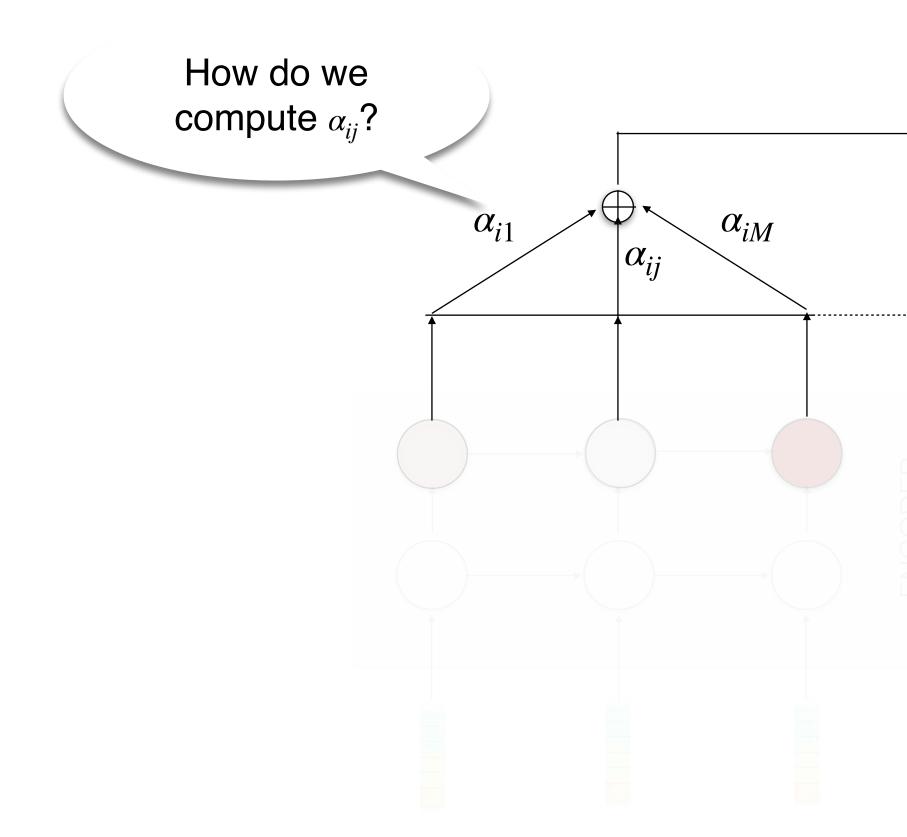


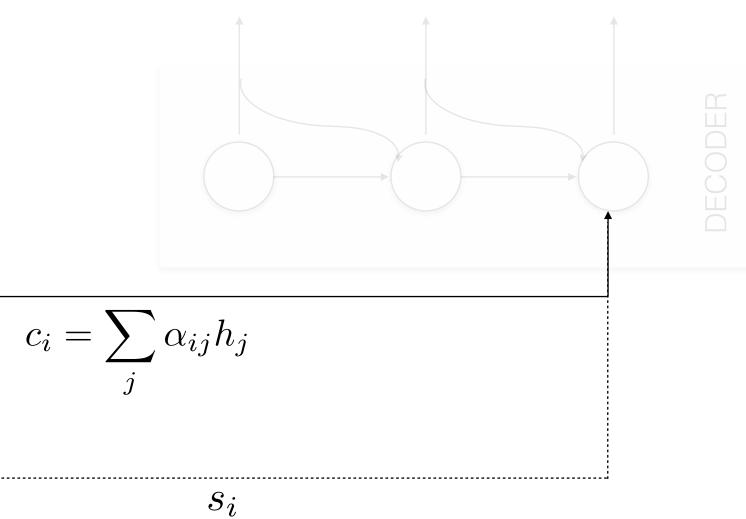


$$c_i = \sum_j \alpha_{ij} h_j$$

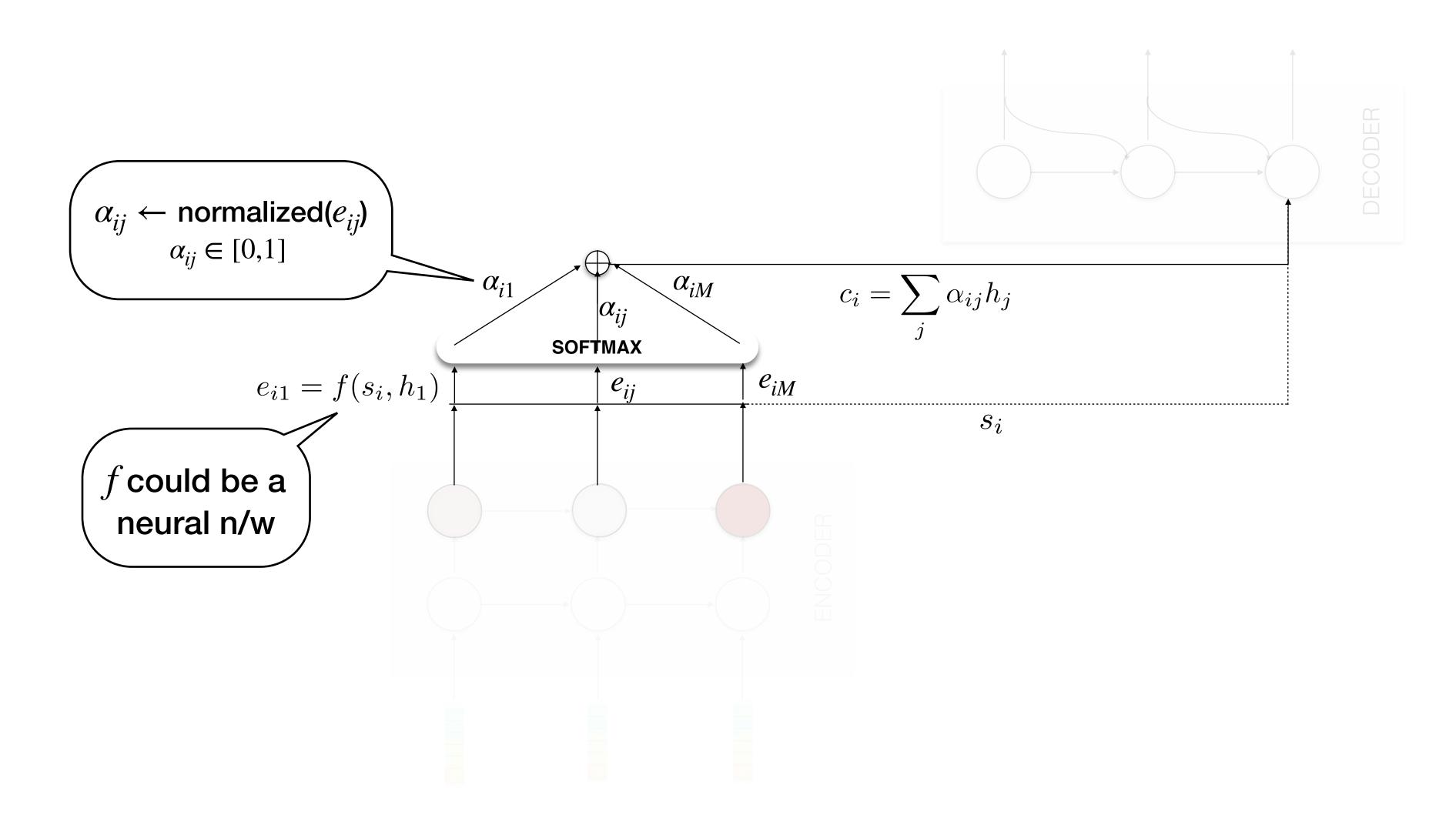
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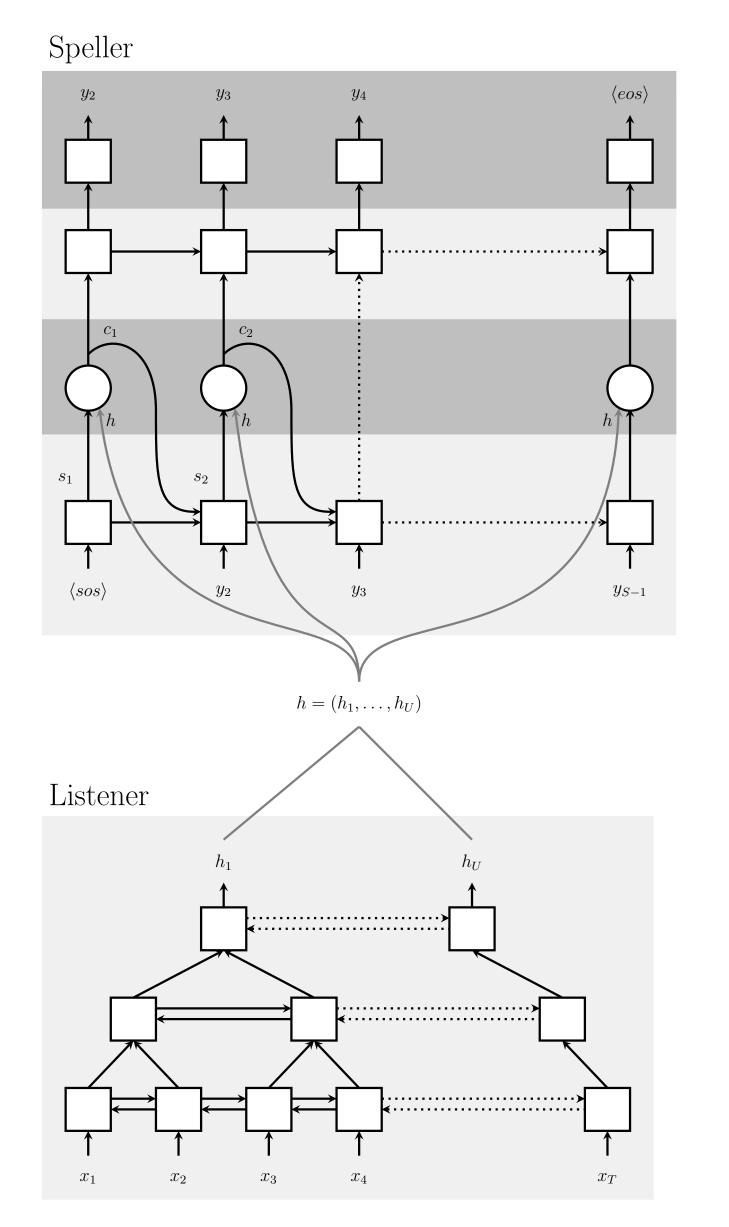
#### **Sequence to sequence models** Encoder-decoder with attention





#### **Sequence to sequence models** Encoder-decoder with attention





### The Model

The Listen, Attend & Spell (LAS) architecture is a sequence-to-sequence model consisting of

- a Listener (Listen): An acoustic model encoder. Deep BLSTMs with a pyramidal structure: reduces the time resolution by a factor of 2 in each layer.
- a Speller (AttendAndSpell): An attentionbased decoder. Consumes **h** and produces a probability distribution over characters.

$$\mathbf{h} = \text{Listen}(\mathbf{x})$$

 $P(y_i | \mathbf{x}, y_{< i}) = \text{AttendAndSpell}(y_{< i}, \mathbf{h})$ 





# **Attend and spell**

- Produces a distribution over characters conditioned on all • characters seen previously
- At each decoder time-step i, AttentionContext computes a score for each encoder step u, which is then converted into softmax probabilities that are linearly combined to compute c<sub>i</sub>

 $e_{i,u} =$ 

- $\alpha_{i,u} =$ 
  - $c_i =$

```
c_i = \text{AttentionContext}(s_i, \mathbf{h})
               s_i = \text{RNN}(s_{i-1}, y_{i-1}, c_{i-1})
P(y_i | \mathbf{x}, y_{< i}) = \text{CharacterDistribution}(s_i, c_i)
```

$$\begin{array}{l} \langle \phi(s_i), \psi(h_u) \rangle \\ \\ \frac{\exp(e_{i,u})}{\sum_{u'} \exp(e_{i,u'})} \\ \\ \\ \sum_{u} \alpha_{i,u} h_u \\ \\ \\ \end{array}$$

## **Training and Decoding**

- Training •
  - probability of the training instances

$$\tilde{\theta} = \max_{\theta} \sum_{i}$$

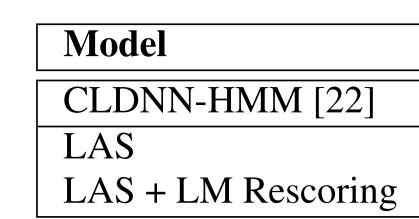
- Decoding ullet
  - Simple left-to-right beam search •
  - Beams can be rescored with a language model

Train the parameters of the model to maximize the log

$$\int \log P(y_i | \mathbf{x}, \tilde{y}_{< i}; \theta)$$

### Experiments

Table 1: WER comparison on the clean and noisy Google voice search task. The CLDNN-HMM system is the state-of-the-art, the Listen, Attend and Spell (LAS) models are decoded with a beam size of 32. Language Model (LM) rescoring can be beneficial.



- used a 2-layer LSTM (256 nodes)
- $\bullet$

Clean WER	Noisy WER
8.0	8.9
14.1	16.5
10.3	12.0

Listen function used 3 layers of BLSTM (512 nodes); AttendAndSpell

Constraining the beam search with a dictionary had no impact on WER

### Analysis

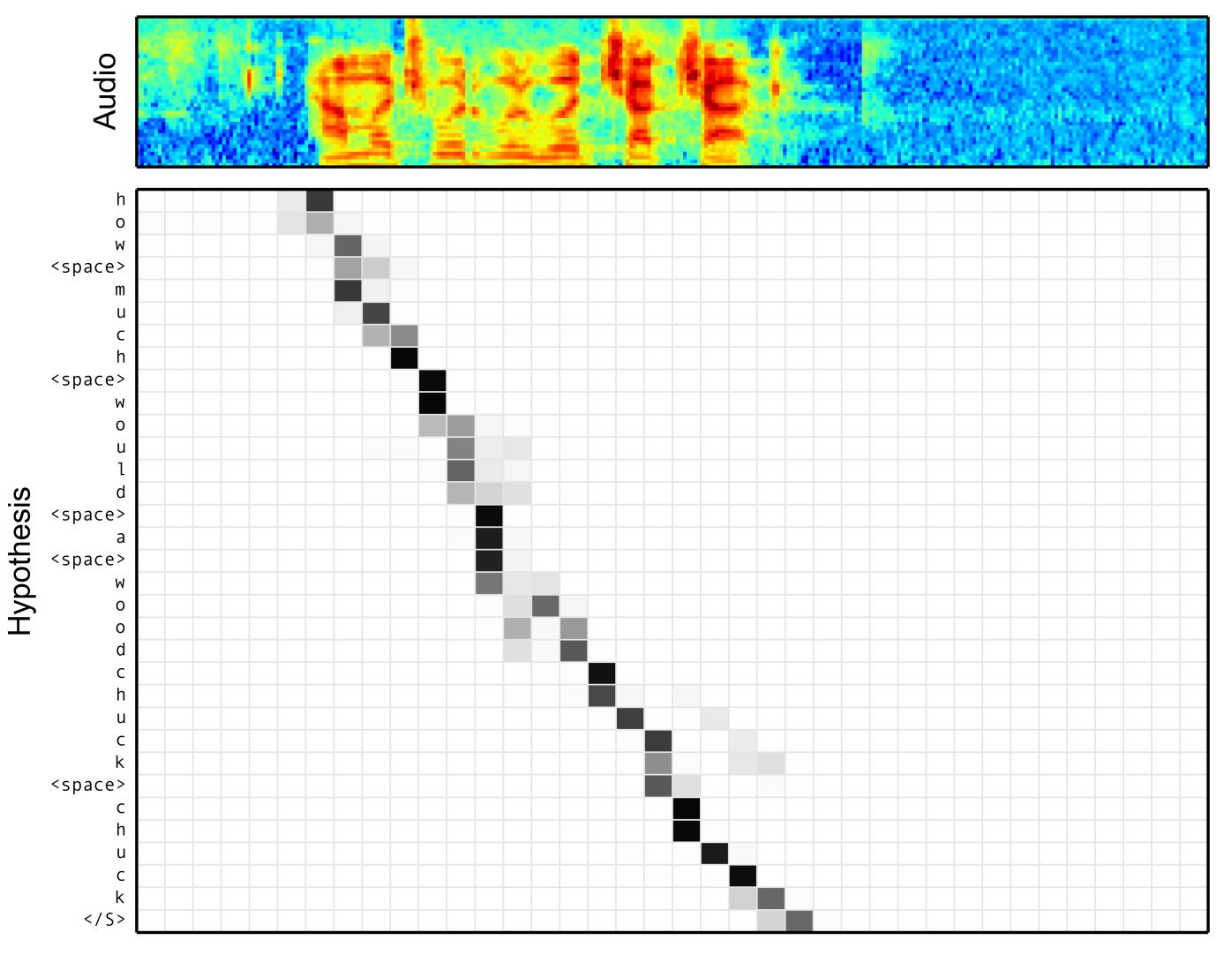


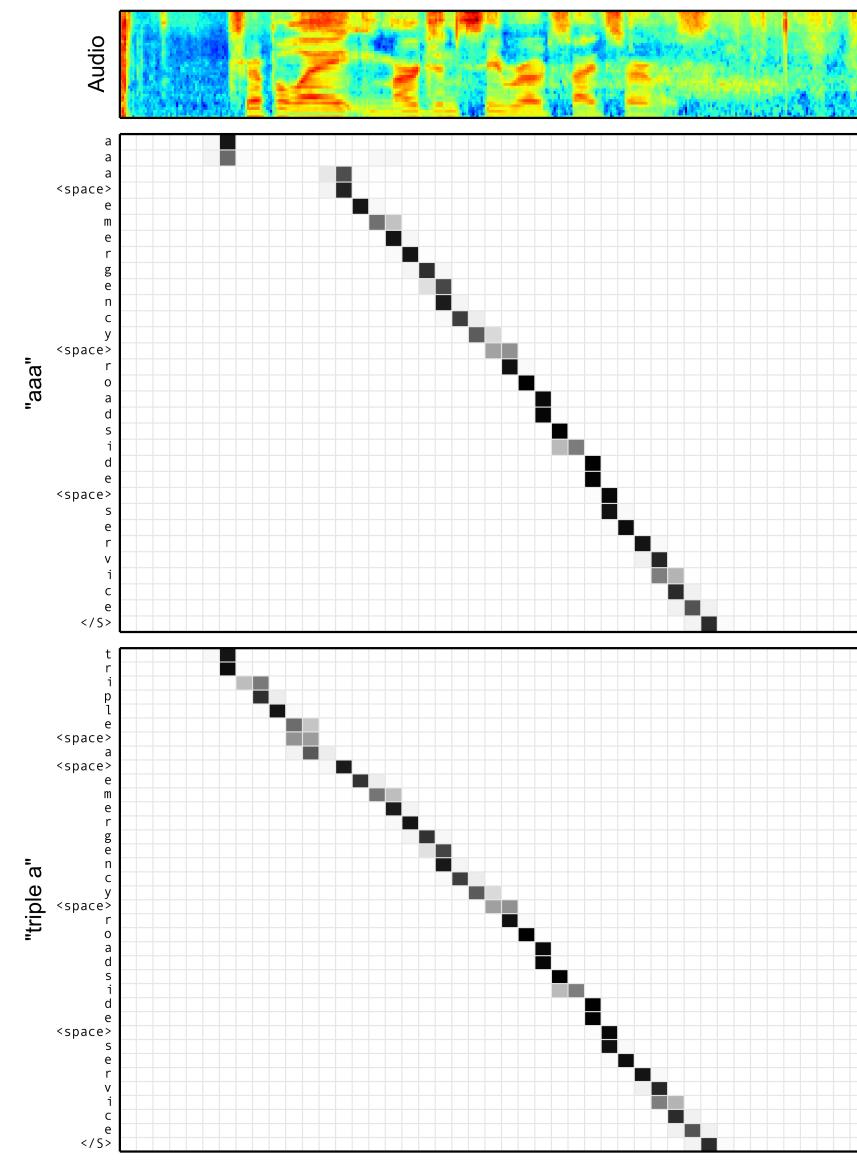
Image from: Chan et al., Listen, Attend and Spell: A NN for LVCSR, ICASSP 2016

Alignment between the Characters and Audio

Time

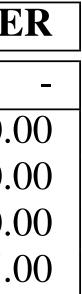
## **Attention Distributions**

#### Spelling Variants of "aaa" vs. "triple a"



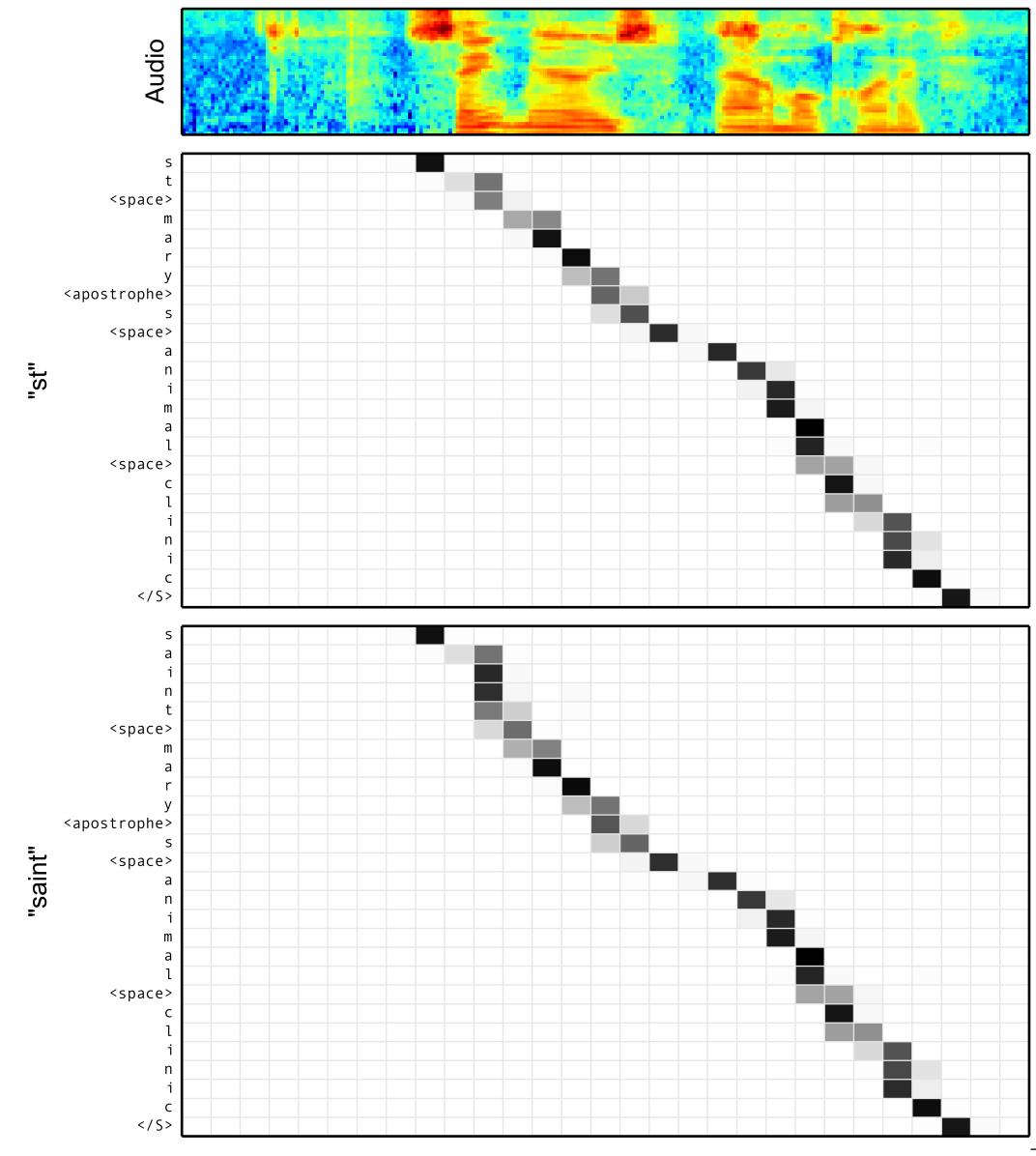


Beam	Text	$\log P$	WE
Truth	call aaa roadside assistance	-	
1	call aaa roadside assistance	-0.57	0.0
2	call triple a roadside assistance	-1.54	50.0
3	call trip way roadside assistance	-3.50	50.0
4	call xxx roadside assistance	-4.44	25.0





## **Attention Distributions**



Spelling Variants of "st" vs. "saint"

