ConvNets for Speech

NYU Lab presentation

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Joint work with Christian Puhrsch Brian Kingsbury Yann LeCun Vaibhava Goel

- Very Deep Multilingual Convolutional Neural Networks for LVCSR http://arxiv.org/abs/1509.08967
- Advances in Very Deep Convolutional Neural Networks for LVCSR http://arxiv.org/abs/1604.01792
- The IBM 2016 English Conversational Telephone Speech Recognition System http://arxiv.org/abs/1604.08242

NN-HMM Hybrid, acoustic model on logmel features

[Abdel-Hamid et al., 2012] [Sainath et al., 2013]



A sloppy picture



XE Cross-Entropy Training



Decoding: getting a WER score



ST Sequence Training



Why CNN is the right acoustic model

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Of course CNNs!

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 - Efficient parametrization
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 - Images: good feature detectors are translation invariant
 - Speech: translation invariance in time, frequency?

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 - Images: good feature detectors are translation invariant
 - Speech: translation invariance in time, frequency?
- ... aren't recurrent networks more powerful?

Computer Vision is not Speech Recognition

ImageNet vs Switchboard

	ImageNet	SWB-1 300h	SWB 2000h	
# frames/images	1.2M	100M	720M	
# classes	1k	8.2k	32k	
image size	224×224	40 × 23		
Class imbalance	No prob	Huge (25% silence)		
Learn Invariance	Viewpoint	Speaker var (Pitch, Accent)		
	Illumination	Structured Noise,		
	Partial obs			

VGG Convolutional Neural Networks

IM . GENET

VGG Convolutional Neural Networks

IM AGENET

- til 2011: Handcrafted + SVM
- 2012: Alexnet: GPUs, ReLU
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[Simonyan and Zisserman, 2014]



2-conv (classic) FC #hmm states FC 2048 FC 2048 FC 2048 FC 2048 4x3 conv. 512 3x1 pool 9x9 conv, 512

input (40x11)

6-conv





10-conv



Result on switchboard

A first look



Multilingual CNN

BABEL - Leveraging many small data sets



Multilingual CNN

BABEL - Leveraging many small data sets



	DNN	1L Clas	6L Clas	1L VC	6L VC
KUR	82.7	82.8	80.6	81.3	78
TOK	62.6	63.3	59.4	59.5	54.3
CEB	76.3	76.7	74.2	73.2	70.6
KAZ	77.3	77.7	75.2	74.4	71
TEL	87.0	86.8	85.4	84.8	82.4
LIT	71.0	72.7	69.5	69.8	66
IMPR	0.00	-0.52	2.10	2.32	5.77

Multiscale Features



Multiscale Features

Results on BABEL



	DNN	3S/20	1S/20	3S/8	1S/8
KUR	82.7	78.1	78.4	78.4	79.2
TOK	62.6	54.2	54.7	55.8	56.7
CEB	76.3	70.3	70.4	71.6	71.8
KAZ	77.3	71.1	71.8	72.5	72.8
TEL	87.0	82.5	83.1	83.5	83.6
LIT	71.0	66.2	67.3	66.9	67.5
IMPR	0.00	5.75	5.20	4.70	4.22

How will we process a full utterance?

Sequence Training, test time



- A: Spliced evaluation, like during Cross-Entropy training
- B: Efficient evaluation

How will we process a full utterance?

Sequence Training, test time



- A: Spliced evaluation, like during Cross-Entropy training
- B: Efficient evaluation possible with any model?



Result on switchboard

Performance hit from architectural constraint



Batch Normalization

- Cross-Entropy training: No Problem. Regular Spatial BatchNorm
- During Sequence Training:
 - Spliced: GPU mem is full with 1 utterance
 - Efficient: stack multiple utts in a batch







Result on switchboard

Getting performance back with Batch Normalization



Result on switchboard

With all bells and whistles (ensemble, big LM)



Training details

- Optimization
 - Fast: Adam + SGD finetuning
 - Better: Pure SGD (with nesterov acceleration)
- Unbalanced data: sample from $p_i = \frac{f_i^{\gamma}}{\sum_j f_j^{\gamma}}$.
- Start from random initialization [-a, a] where $a = (kW \times kH \times numInputFeatureMaps)^{-\frac{1}{2}}$.

Objective mismatch

Some objectives we don't care about

- Frame-level cross-entropy $\mathcal{L}_u = -\sum_t \log y_{ut}(s_{ut})$
- CTC for E2E training of RNNs $\mathcal{L}_u = \sum_{\pi \in B^{-1}(I_u)} \prod_{t=1}^{T_u} y_{\pi_t}^t$

• Expected Sentence Error, e.g. MMI: $\mathcal{L}_u = \log \frac{p(X|S_u)P(W_u)}{\sum_w p(X|S)P(W)}$

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What we do care about

• Word Error Rate

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• Word Error Rate (?) - to publish papers

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What we do care about

- Word Error Rate (?) to publish papers
- Real life usability
 - Certain words are more important: weighted word error rate?
 - Segmentation into utterances, silence detection
 - Domain mismatch: noise, accents

Objective mismatch - How well aligned are XE and WER?



 ${\sf E}$ van den Berg, B Ramabhadran, M Picheny, "Neural network training variance and performance evaluation in speech"

• Expect filters to be sensitive to certain frequency regions?

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- Help the filters to be sensitive to certain frequency regions
 - Bias per-frequency
 - ... or batchnorm statistics per (featuremap, frequency)

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Conclusion

Overview

Very deep convolutional networks

- Small 3×3 kernels
- Multiple convs before pooling
- Best arch: 10 convs, 14 total
- 10.6% improvement over classic CNNs (300h, CE)

Multilingual training

Shared convolutional layers

Multiscale features

Same computation, more context