

intent-bear

AIRBUS ATC challenge



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intent-bear ... WHO ARE WE?

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intent-bear ... WHY AIRBUS ATC CHALLENGE?

- experience from the IT-BLP project
 - Intelligent technologies for improving air traffic security
 - Supported by GAČR 2011-2015
- cooperation UWB & JHU
 - JHU - CLSP - KALDI developer

IT-BLP project

Tasks:

- Collect, process, and transcribe approx. 200 h of recordings from ANS/RLP Praha
- ASR (web demonstrator using technologies WebRTC, SIP, WebSockets, LVCSR, Tornado, Python)
- TTS - specific voices with accent (Czech, British, American, Serbian, German, Polish, France, Chinese, ...)
- aTT - automatic training tool (video: goo.gl/zn6kU8)
 - Web application for creating teaching/learning material
- aPP - automatic pseudo-pilot (video: <https://goo.gl/JwdJCv>)
 - Multimodal dialog system designed as a learning tool for air traffic control officer trainees (ATCO)

Demo & technological demonstrator (year 2015):

- itblp.zcu.cz/



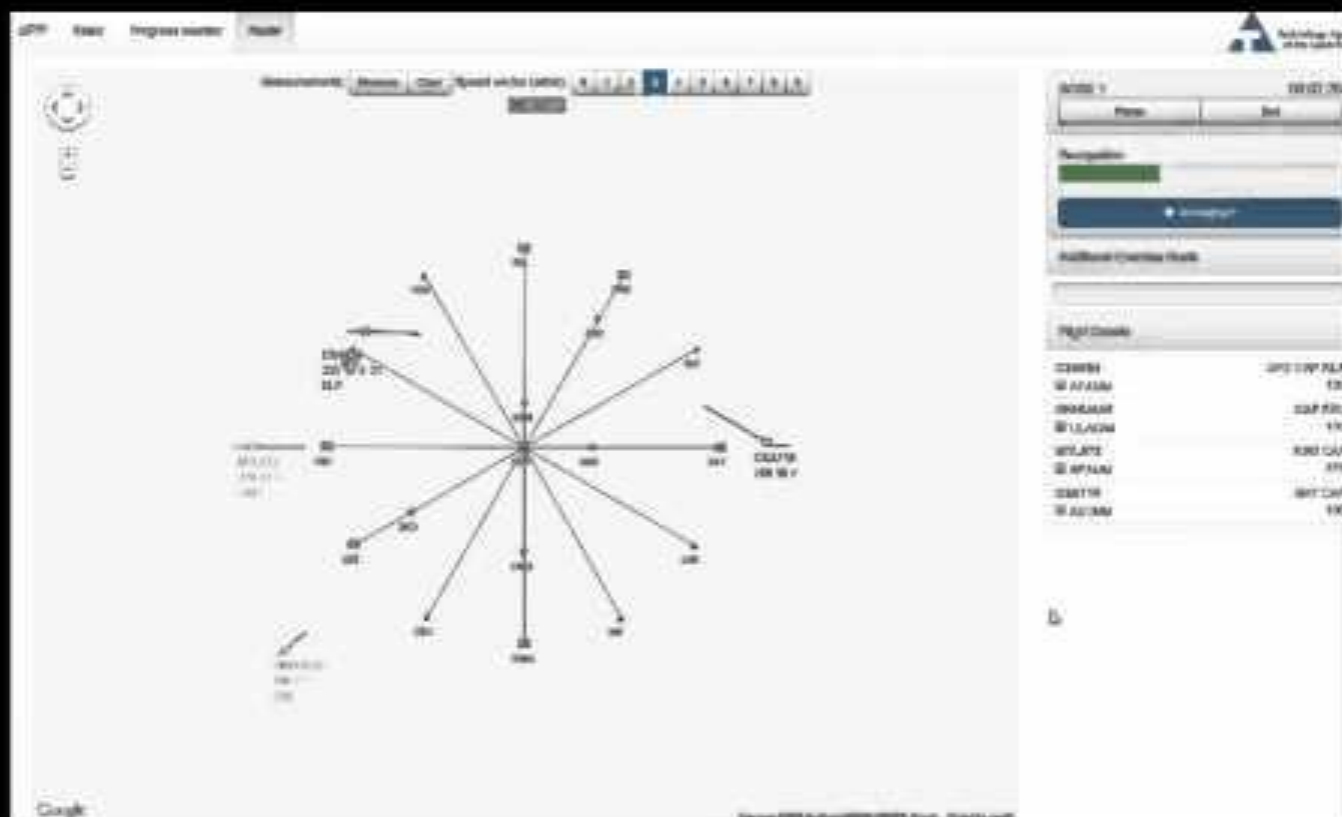
APP - Automatic Pseudo-Pilot

A multimodal dialogue system for ATC trainees

Functionality:

- understand ATC's utterance (ASR+SLU) + answer (TTS +noises)
- control air traffic generator - ATG
- show simulated radar screen - HTML5
- GUI of the dialogue system
- shows output of ATG, connects to ASR and TTS
- evaluate user's performance
- recorded radar screen with timeline of user's actions
- flight statistics for each airplane
- create different situation to exercise
- assign flight plans and additional goals

APP



ATC Challenge

Leaderboard results:

- 2nd place (harm mean: 0.98)

Test results:

- 4th place (WER 0.0876, F1 0,7704)

Footnotes:

- We have enjoyed it
- We can do more - another improvement after the competition ...
- We are able to train production ASR for a different location (semi-supervised)

ASR overview

- KALDI-based ASR
- Deployment-ready single system

Overview

- Lexicon preparation
- Language modeling
- Additional data?
- Handling <UNK> and <FOREIGN> tokens

Lexicon preparation

Out of 2500 types in the training list, around 500 were typos.

We checked against CMUdict

- Fixed manually typos
- Generated french pronunciation for french words (cities) using espeak + manually created table IPA->ARPAbet
- Verified specific words do exist (ATC terminology, waypoints)
- Trained G2P for correct, words not present CMUdict (phonetisaurus)
- Added 'huh' pronunciations (from WSJ)
- Two possible <UNK>: unknown word '_' and foreign word (or phrase) '@'



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Language Modeling

Used srilm toolbox

- 3-gram perplexity: 8.0
- 4-gram perplexity: 5.0 (MaxEnt LM, used for rescoring)

RNNLM didn't help

No other external data

Additional data available ?

Youtube channels (approx 100 hrs recordings)

LiveATC

- Fan-driven community page containing recordings of communication from various airports
- downloaded around 150k hours of recordings
- FR, CZE, SW, US, CAN accents

UWB corpus (proprietary corpus of approx 200 hrs of CZE accented ATC - IT-BLP data)

Various sites with additional aux info: phraseology, spelling, aviation-safety, manuals, plane crashinfo, quora, skytalk, tailstrike



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Handling the <UNK>

- Typically, detecting UNKs is fairly hard task
- Normally, you'd see something like this in a lexicon
 - <UNK> <unk>
 - I.e. word '<UNK>' maps to a single unit '<unk>'
 - This way, the training procedure is able to use the sentence for training, but the model of '<UNK>' won't be very good
- For decoding, it is a better idea to replace the pronunciation of '<UNK>' by a phoneme graph
 - Either all probabilities constant
 - Or you can train a LM on alignment of the training data

```
48
49 if [ $stage -le 4 ] ; then
50   utils/lang/make_unk_lm.sh data/local/dict exp/make_unk
51
52   utils/prepare_lang.sh \
53     --unk-fst exp/make_unk/unk_fst.txt --phone-symbol-table data/lang/phones.txt \
54     data/local/dict "<UNK>" data/local/lang_test data/lang_test
55
```

Handling the <UNK>

First idea: map both '_' and '@' to <UNK>

Second idea from listening to audio: map '@' to <FOREIGN> with pronunciations of French greetings.

```
637 <FOREIGN> b oh n jh uw r ah
638 <FOREIGN> b ah n sh uh r
639 <FOREIGN> b oh n jh uw r
640 <FOREIGN> b aa n
641 <FOREIGN> b oh n
642 <FOREIGN> jh uh r n ea
643 <FOREIGN> ow r ah v w aa
644 <FOREIGN> ow r ah v w aa r
645 <FOREIGN> oh r eh v uh aa r
```

```
17 if [ $stage -le 4 ] ; then
18   utils/prepare_lang.sh \
19     --unk-fst exp/make_unk/unk_fst.txt --phone-symbol-table data/lang/phones.txt \
20     data/local/dict_foreign/ "<UNK>" data/local/lang_foreign_test data/lang_foreign_test
21
22   utils/format_lm.sh \
23     data/lang_foreign_test data/srilm_foreign/best_3gram.gz data/local/dict_foreign/lexicon.txt data/lang_foreign_test
24
25   utils/build_const_arpa_lm.sh \
26     data/srilm_foreign/best_4gram.gz data/lang_foreign_test data/lang_foreign_test_fg
```

Adding <FOREIGN>

- Hypothesis easy to test -- generate new lexicon and decoding graph, decode again
 - Make sure you use the '--phone-symbol-table' parameter for make_lang.sh
- Can we train? Remember not all <FOREIGN> can be salutations
 - Yes, we can
 - Utterances that fail the alignment will get removed automatically
- Too many utterances dropped? Add line
 <FOREIGN> <foreign>
Into the lexicon (and into phone list)
We tried this and for this case it made results worse



Pronunciation probabilities

- Most lexicons do not specify which pronunciation variant is more probable.
- For some words, the silence is more probable than after other (this probability is not modeled by LM)
- We can use our alignments to estimate these probabilities
- In practice, the conditional silence probability seems to be more important

PRONUNCIATION AND SILENCE PROBABILITY MODELING FOR ASR

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Pronunciation probabilities

```
228  
229 steps/get_prons.sh --cmd "$train_cmd" data/train_nodup data/lang_nosp exp/tri3b  
230  
231 utils/dict_dir_add_pronprobs.sh --max-normalize true \  
232   data/local/dict_nosp exp/tri3b/pron_counts_nowb.txt exp/tri3b/sil_counts_nowb.txt \  
233   exp/tri3b/pron_bigram_counts_nowb.txt data/local/dict  
234  
235 utils/prepare_lang.sh data/local/dict "<unk>" data/local/lang data/lang
```

1. First, get the stats from the alignments (of the training data)
2. Create a new dict dir
3. Generate lang directory the usual way
4. Add G.fst and regenerate decoding graph (not shown)

Data cleanup

- The transcribed data will often contain transcription errors, the segments are not correct, audio can be so noisy, that it causes harm using it...
- Idea: recognize using biased LM and use only those parts that were recognized correctly
- Used fairly often in kaldι egs
- Typically done before DNN training to get nice/correct alignments
- Script local/run_cleanup_segmentation.sh

```
45 # This does the actual data cleanup.
46 steps/cleanup/clean_and_segment_data.sh --stage $cleanup_stage \
47     --nj $nj --cmd "$cmd" \
48     $data $langdir $srcdir $dir $cleaned_data
```

Acoustic model

- Chain model (LF-MMI), factorized TDNN
- 12-layer, dim=1280, bottleneck=256, dropout
- Unconstrained egs
- Data cleanup (10 % of the data thrown away)
- Data augmentation: volume and speed (final system had 5-way, but performed only marginally better than “standard” 3-way)
- i-vectors (fairly small gain), tested two-pass i-vector estimation, again very tiny gain
- UNK = 4-gram phoneme loop
- Online decoder

Internal Results (train split into 30+5(dev)+5(test))

Baseline 9.28

+ Cleanup	9.02
+ iVectors	8.98
+ Pronprobs	8.83
+ LM Rescoring	8.45
+ <FOREIGN>	7.69
+ Two-stage ivectors	~0.03 (not included)
+ 7-way augmentation	~0.00



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ASR submissions details

Three different submissions

- Single system, TDNN -- driven by our philosophy, that the competing submission should reflect deployable solution -- real-time decoder, no (many)system combination, no (B)LSTM
- Three different submissions had the same AM two LMs
 - <FOREIGN> mapped to <UNK>
 - <FOREIGN> modelled as French phrases
- For Eval run, we have included dev and test, i.e. we trained on 40 hrs of speech. This gave 0.3 % improvement on leaderboard data.

Call-sign detection - initial experiments

Reuse the semantic entity detection method from IT-BLP project

Many drawbacks in the challenge:

- Designed to work with ASR lattices
- Outputs the unified description of entity
- Uses expert-defined context-free grammars

Advantages not usable in the challenge:

- Allows to sum-up multiple ASR hypotheses with the same meaning
- Multiple output hypotheses with posterior scores

Call-sign detection - trainable model

2-layer bidirectional LSTM

- Training data
 - Recognized ASR hypothesis with ground truth callsign (alignment!)
 - Transcribed train partition
 - Recognized train partition
 - Recognized dev & test partitions
- LSTM tagging
 - Output classes: no CS, beginning of CS, middle of CS, end of CS
- Expert knowledge (word classes) by additional embedding layer
 - Company name, numbers, spelling alphabet
- Ensembling to average over different initializations of LSTM training

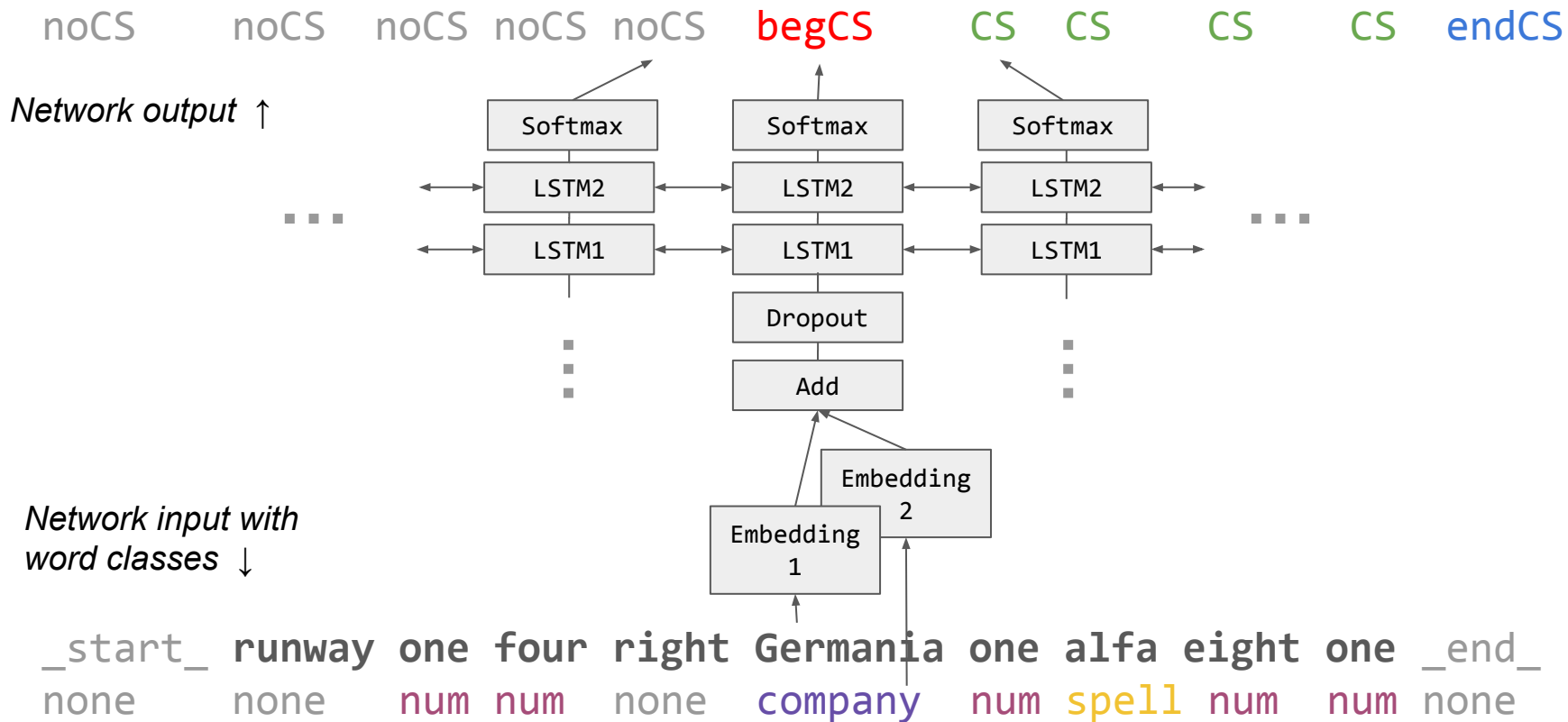


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Network architecture



Submissions details

We are using different LMW & WIP weights for ASR submission and CS detection

- Optimized on dev data
- Typically, the CS detection performs better with higher LMW

LSTM ensemble (3-5 averaged networks)

- to minimize the noise from different LSTM initializations

Improvement in WER \nRightarrow improvement in F1

- esp. for our train/dev/test split and leaderboard data

Call-sign detection results

F1 metrics on leaderboard data

- | | |
|---|---------------|
| • Semantic entity detection (expert-based) | 0.7021 |
| • Initial experiment with LSTM (1 LSTM layer) | 0.7984 |
| • Full-featured LSTM model | 0.8340 |

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