

# A toolkit for speech recognition research

(According to legend, Kaldi was the Ethiopian goatherd who discovered the coffee plant).



## Key aspects of the project

- Apache v2.0 license (very free)
- Available on Sourceforge
- Open source, collaborative project (we welcome new participants)
- C++ toolkit (compiles on Windows and common UNIX platforms)
- Has documentation and example scripts



#### Overview of features

- Context-dependent LVCSR system (arbitrary phonetic-context width)
- FST-based training and decoding (we use OpenFst)
- Maximum Likelihood training
  - Working on lattice generation + DT.
- All kinds of linear and affine transforms
- Example scripts demonstrate VTLN, SAT, etc.



## Advantages versus other toolkits\*

- Clean code, modular and extensible design
- Intended to be easy to understand and modify
- Very open license (Apache 2.0)
- Example scripts and documentation available
- Trying to build helpful and active community
- Good linear algebra support; FSTs
- Very scalable
- We intend to implement all state-of-the-art methods (inc. discriminative training)

<sup>\*</sup>Disclaimer: some toolkits may have at least some of these advantages.

#### Features not on current "to-do" list

- No on-line decoder (batch mode only)
  - It's mostly for speech-recognition research
- No explicit support for parallelization (MapReduce, MPI)
  - Would be too platform-specific... we use a HTKlike approach where you can sum up accumulator files.
- No scripting-language wrapper
  - Would force users to learn e.g. Python; we support configurability in different ways.
- No forward-backward training
  - We don't believe it's better than Viterbi; and Viterbi makes it convenient to write alignments to disk.

# History of Kaldi (1/2)

- JHU 2009 workshop\*, working (mostly) on Subspace Gaussian Mixture Models (SGMMs)
- Guys from Brno University of Technology (BUT) created "proto-Kaldi"...
  - Had FST-based decoding and SGMM training setup
  - Dependent on HTK for feature generation and building an initial GMM-based system
  - Entire solution was complex due to merging of two different setups.

\*This workshop was funded by NSF Grant #IIS-0833652, with supplemental funding from Google, DARPA's GALE program, and JHU's HLTCoE. BUT researchers were partially supported in this period by Czech Ministry of Trade and Commerce Project # FR-TI1/034, Grant Agency of Czech Republic project no. 102/08/0707, and Gzech Ministry of Education project no. MSM0021630528. Arnab Ghoshal was partially supported during this period by the European Community's Seventh Framework Programme under grant agreement number 213850 (SCALE).

# History of Kaldi (2/2)

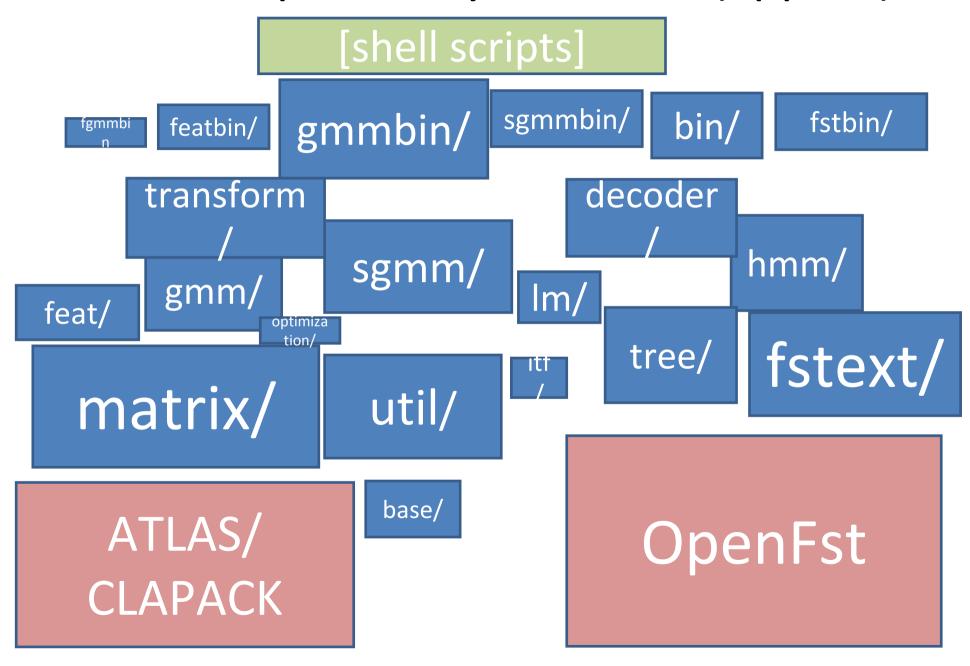
- In summer of 2010, some of us (+ new participants) went back to Brno for 2 months ("Kaldi workshop 2010"), hosted by Brno University of Technology.
- Aimed to build a self-contained, clean toolkit with no HTK dependency.
- Immediate goal was to create clean, releasable SGMM recipe.
- Wider goal of making a clean speechrecognition toolkit.
- Completed a lot of it that summer but not ready for release until last week.

## Kaldi contributors

- Individuals who wrote code for Kaldi\* so far:
  - Mohit Agarwal, Sandeep Boda<sup>1</sup>, Gilles Boulianne, Lukas Burget, Arnab Ghoshal, Mirko Hannemann, Ondrej Glembek, Nagendra Goel<sup>1</sup>, Pavel Matejka<sup>2</sup>, Petr Motlicek, Daniel Povey<sup>3</sup>, Yanmin Qian, Ariya Rastrow, Sandeep Reddy<sup>1</sup>, Petr Schwarz<sup>2</sup>, Jan Silovsky, Georg Stemmer, Karel Vesely, Haihua Xu.
- Also thanks to (non-exclusively)\*\*:
  - Alex Acero, Pinar Akyazi, Honza Cernocky, Paul Dixon, JHU's CLSP staff + faculty, Tomas Kasparek, Renata Kohlova, Rico Malvar, Patrick Nguyen, Mike Riley, Rick Rose, Samuel Thomas, Geoffrey Zweig.

<sup>1</sup>GoVivace, Inc. <sup>2</sup>Phonexia s.r.o. <sup>3</sup>Microsoft Corp. (code contributed as employee)
\*I.e. specifically for Kaldi \*\*There are probably inadverted by ensights.

## Kaldi dependency structure (approx)



## Matrix library

- C++ wrapper for BLAS and CLAPACK linear algebra libraries (plus some extra code).
- Can use either (BLAS+CLAPACK), or ATLAS, or MKL, as external library.
- Supports generic, packed symmetric and packed triangular matrix formats.
- Supplies typical linear-algebra functionality (SVD, etc.), and FFT.
- Reusable: independent of rest of Kaldi code (except one small directory "base/").

# OpenFst and fstext/

- OpenFst is open-source FST library (mostly from Google)
- We compile against it, e.g. decoding-graph object is an OpenFst object.
- fstext/ contains various extensions to OpenFst
  - E.g. implementation of on-demand contextdependency transducer
  - Our FST recipe is a little bit different from the standard one and requires slightly different FST algorithms (e.g. determinization with epsilon removal)

## Kaldi I/O

- Based on C++ streams
- Supports binary and text-mode formats
- extended filenames: "-", "gunzip -c foo.gz|",
   "/offset/into/file:12345"
- Archive format: generic mechanism to index objects by strings (typically utterance id)



## Tree building and clustering code

- Very generic clustering and tree building mechanisms
- Easy to build trees in various different ways (globally shared tree roots, etc.)
- Our current recipes use automatically generated questions (minimize hassle)
- Mechanisms scalable to wide context (e.g. quinphone) and large phone-sets
- In WSJ recipe we, in effect, ask questions about phone-position and stress (via expanded phone set and specially constrained questions... this is mostly set up at the script level)

## HMM and transition modeling

- This code is separate from the "GMM" side of things (just treat states as integer ids)
- Can specify a "prototype" topology for each phone
- Transition is separately estimated depending on the p.d.f. index on the state it comes out of
- Mechanisms for turning these HMMs into FSTs
- In our FSTs, the (input) labels encode more information than just the p.d.f. index (e.g. encodes the phone, the position in the HMM)
- This is so we can train the transitions (and can work out the phone sequences from this index sequence)



## Decoding-graph creation

- There is a C++ mechanism for creating decoding graphs (FSTs) in training time, from transcriptions
- These graphs are typically cached on disk
- We train using the Viterbi path through these graphs (redo Viterbi every few iterations)
- For the larger decoding graphs used in test time, we put relatively simple command-line tools together with a shell script
- Some of these are OpenFst tools, but mostly our own (using C++-level OpenFst mechanisms),

## Gaussian Mixture Models (GMMs)

- Code for GMMs is fairly simple and passive
- Have avoided complex frameworks
- Representation of a single GMM
  - Likelihood evaluation; mutators
  - Separate class for accumulation and training
- Class for a collection of GMMs (indexed by integer pdf-index)... similar to vector<Gmm>
  - Corresponding "accumulator" class
- GMM code does not "know about" HMMs, transition models, linear transforms, etc.

## Linear transform code

- Code for estimation of various linear transforms
  - LDA, HLDA, fMLLR/CMLLR, MLLT/STC, linear VTLN, "exponential transform" (something new, like VTLN), MLLR
  - This code is specifically for GMMs (would code these algorithms separately for other models)
- Linear transforms applied in a unified way (code does not "know" how they were estimated)
  - Usually applied as part of a pipe
- Mechanisms for regression trees for (fMLLR, MLLR)
  - Used in separate command-line decoders (don't want to complicate code that isn't doing this)

### Decoders

- Decoders (currently) use fully expanded FSTs
- Currently 3 decoders on spectrum simple ←→fast
- But >3 command-line decoding programs!
- Decoders don't "know about" GMMs, HMMs, etc: just the FSTs, and "Decodable" interface
- "Decodable" interface has function that says "give me score for this (frame, index)"... like matrix lookup
- We "wrap" GMMs etc. in a thin wrapper that satisfies "Decodable" interface
- Command-line decoding programs always do one pass of decoding and are for a specific (decoder, model type).
- Multiple decoding passes done at script level (invoke decoder multiple times)

## Feature processing

- Support standard MFCC and PLP features
- A reasonable range of configurability (#mel bins, etc.)
- Read only .wav format
- Use external programs for format conversion,
   e.g. from sphere
- Typically write features (like other objects) all to a very large file
- Expansion with deltas, fMLLR, etc. typically done using pipes, on-the-fly, to minimize disk I/O
- In next talk will explain the framework for this

### Command-line tools

- Large number of command-line tools (>150), each with a fairly simple function
- Command-line tools take options e.g.

```
compute-mfcc-feats --use-energy=false \
    ark:data/train_wav.scp \
    ark,scp:data/train.ark,train,scp
```

- We rarely need to supply more than a few options to any given program
- Command line tools generally have quite simple code.
- C++ code doesn't have to worry much about I/O (handled through templated code via "Table" concept... will explain after the break) KALDI

# Scripts (example fragment)

```
#!/bin/bash
while [$x -lt $numiters]; do
if echo $mllt iters | grep -w $x >/dev/null; then # Do MLLT update.
  (ali-to-post ark:$dir/cur.ali ark:- | \
    weight-silence-post 0.0 $silphonelist $dir/$x.mdl ark:- ark:- | \
    gmm-acc-mllt --binary=false $dir/$x.mdl "$featsub" ark:- $dir/$x.macc ) \
     2> $dir/macc.$x.log || exit 1;
   est-mllt $dir/$x.mat.new $dir/$x.macc 2> $dir/mupdate.$x.log || exit 1;
   gmm-transform-means --binary=false $dir/$x.mat.new $dir/$x.mdl $dir/$[$x+1].mdl \
   2> $dir/transform means.$x.log || exit 1;
   compose-transforms --print-args=false $dir/$x.mat.new $cur | Ida $dir/$x.mat | | exit 1;
   cur_lda=$dir/$x.mat
  feats="ark:splice-feats scp:data/train.scp ark:- | transform-feats $cur | Ida ark:- ark:- | "
   # Subset of features used to train MLLT transforms.
   featsub="ark:scripts/subset_scp.pl 800 data/train.scp | splice-feats scp:- ark:- |
       transform-feats $cur_lda ark:- ark:- | "
else
```

## Scripts (points to note)

- Scripts quite complex
- Much of the configurability of Kaldi takes place at shell-script level
- This helps keep the C++ code simple
- Note use of pipes: features, alignments etc. are passed through pipes.



# Selected results (WSJ)

%WER	Nov'92	Nov'93
Reichl (2000)	11.9	15.4
HTK (gender dep.) (ICASSP'94)	11.1	14.5
Kaldi	11.8	15.0

- SI-284 training, Sennheiser microphone, 20k open vocabulary test, bigram LM supplied with WSJ.
- Unadapted, cross-word triphones (but HTK system was gender-dependent)
- This is not our best result, just showing that with comparable algorithms we get comparable results)

# Speed, decoding issues (WSJ)

- We can't yet decode with full trigram LM from WSJ (graph too large)... but pruned one is OK
- Working on this issue
- Decoding speed for previous results is about 0.5xRT (i.e. twice faster than real time)
- Training time: takes a few hours to train the previous system, on a single machine (using up to 3 CPUs)



# Further results (RM)

%WER	Feb'89	Oct'89	Feb'91	Sep'92	Avg
HTK	2.77	4.02	3.30	6.29	4.10
Kaldi	3.20	4.10	2.86	6.06	4.06

- Both systems cross-word triphone with cepstral mean normalization
- HTK results from ICASSP'99 paper (Povey et. al)
  - Probably slightly better than RMHTK recipe due to variable #gauss per state.
- Decoding speed ~0.1xRT



## RM: unadapted experiments

%WER	None	
$MFCC+\Delta + \Delta\Delta$	4.59	
+mean normalization	4.16	
$MFCC+\Delta+\Delta\Delta$ + $MLLT/STC$	4.59	
Splice-9 + LDA	5.04	
Splice-9 + LDA + MLLT/STC	4.35	
Splice-9 + HLDA	4.61	
Triple-deltas + HLDA	4.32	
Triple-deltas + LDA + MLLT/STC	3.95	
SGMM	3.15	

All results averaged over 6 RM test sets LDI

## RM: adapted experiments

%WER	Utt	Spk
MFCC+ $\Delta$ + $\Delta\Delta$ (fMLLR)	4.56	3.67
$MFCC+\Delta + \Delta\Delta + ET$	3.35	3.32
$MFCC+\Delta + \Delta\Delta + VTLN$	3.94	3.56
Splice-9 + LDA + ET	3.29	3.08
+ fMLLR		2.73
Splice-9 + LDA + MLLT/STC + SAT	5.10	2.75
SGMM + spk-vecs	2.72	2.68
SGMM + spk-vecs + fMLLR		2.53
SGMM + fMLLR		2.77

• "ET" = Exponential Transform (like TKIX) LDI

## Why use Kaldi

- Easy to use (once you learn the basics, and assuming you understand the underlying science)
- Easy to extend and modify.
- Redistributable: unrestrictive license, community project.
- If you stuff works or is interesting, the Kaldi team is open to including it and your example scripts in our central repository -> more citations, as others build on it.