Introduction to HTK Toolkit



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Reference:

- Steve Young et al. The HTK Book. Version 3.4, March 2006.

Outline

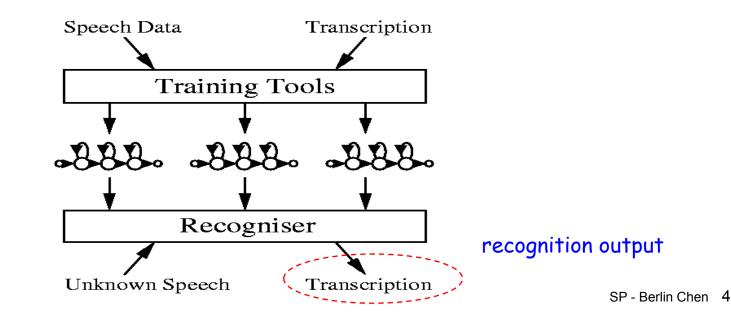
- An Overview of HTK
- HTK Processing Stages
- Data Preparation Tools
- Training Tools
- Testing Tools
- Analysis Tools
- Homework: Exercises on HTK

An Overview of HTK

- HTK: A toolkit for building Hidden Markov Models
- HMMs can be used to model any time series and the core of HTK is similarly general-purpose
- HTK is primarily designed for building HMM-based speech processing tools, in particular speech recognizers

An Overview of HTK (cont.)

- Two major processing stages involved in HTK
 - Training Phase: The training tools are used to estimate the parameters of a set of HMMs using training utterances and their associated transcriptions
 - Recognition Phase: Unknown utterances are transcribed using the HTK recognition tools

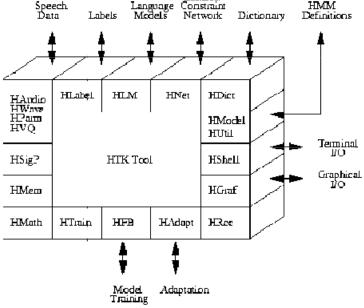


An Overview of HTK (cont.)

- HTK Software Architecture
 - Much of the functionality of HTK is built into the library modules
 - Ensure that every tool interfaces to the outside world in exactly the same way
- Generic Properties of an HTK Tools
 - HTK tools are designed to run with a traditional command line style interface
 Speech Language Constant Medels Network Dictional

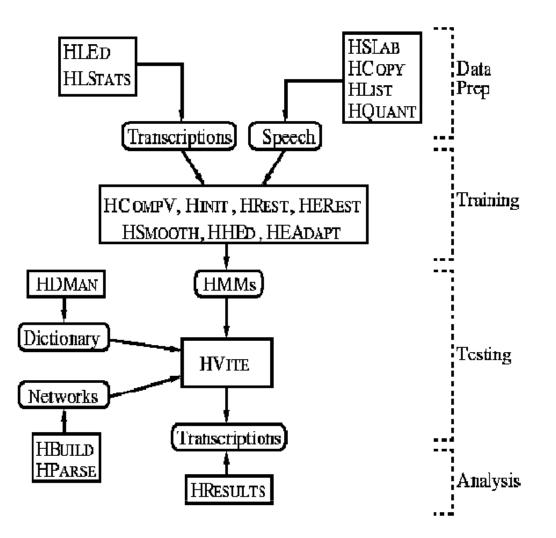
HFoo -T -C Config1 -f 34.3 -a -s myfile file1 file2

 The main use of configuration files is to control the detailed behavior of the library modules on which all HTK tools depend



HTK Processing Stages

- Data Preparation
- Training
- Testing/Recognition
- Analysis



Data Preparation Phase

- In order to build a set of HMMs for acoustic modeling, a set of speech data files and their associated transcriptions are required
 - Convert the speech data files into an appropriate parametric format (or the appropriate acoustic feature format)
 - Convert the associated transcriptions of the speech data files into an appropriate format which consists of the required phone or word labels
- HSLAB
 - Used both to record the speech and to manually annotate it with any required transcriptions if the speech needs to be recorded or its transcriptions need to be built or modified

Data Preparation Phase (cont.)

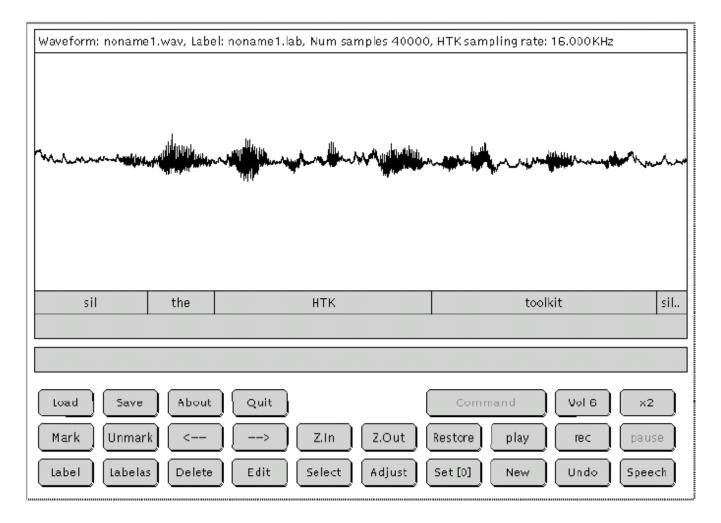
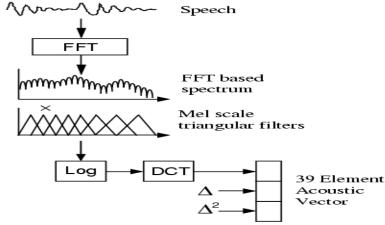


Fig. 14.1 HSLab display window

Data Preparation Phase (cont.)

- HCOPY
 - Used to parameterize the speech waveforms to a variety of acoustic feature formats by setting the appropriate configuration variables



MFCC-based Front-End Processor.

LPC	linear prediction filter coefficients	
LPCREFC	linear prediction reflection coefficients	
LPCEPSTRA	LPC cepstral coefficients	
LPDELCEP	LPC cepstra plus delta coefficients	
MFCC	mel-frequency cepstral coefficients	
MELSPEC	linear mel-filter bank channel outputs	
DISCRETE	vector quantized data	

Data Preparation Phase (cont.)

• HLIST

 Used to check the contents of any speech file as well as the results of any conversions before processing large quantities of speech data

• HLED

 A script-driven text editor used to make the required transformations to label files, for example, the generation of context-dependent label files

• HLSTATS

- Used to gather and display statistical information for the label files

• HQUANT

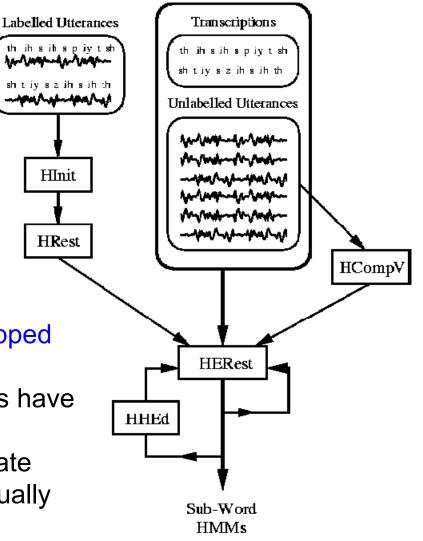
 Used to build a VQ codebook in preparation for build discrete probability HMM systems

Training Phase

- Prototype HMMs
 - Define the topology required for each HMM by writing a prototype Definition
 - HTK allows HMMs to be built with any desired topology
 - HMM definitions stored as simple text files
 - All of the HMM parameters (the means and variances of Gaussian distributions) given in the prototype definition are ignored only with exception of the transition probability

```
~o <VecSize> 39 <MFCC_0_D_A>
"h "proto"
<BeginHMM>
 <NumStates> 5
 <State> 2
    <Mean> 39
      0.0 0.0 0.0 ...
    <Variance> 39
      1.0 1.0 1.0 ...
 <State> 3
    <Mean> 39
      0.0 0.0 0.0 ...
    <Variance> 39
      1.0 1.0 1.0 ...
 <State> 4
    <Mean> 39
      0.0 0.0 0.0 ...
    <Variance> 39
      1.0 1.0 1.0 ...
 <TransP> 5
 0.0 1.0 0.0 0.0 0.0
 0.0 0.6 0.4 0.0 0.0
  0.0 0.0 0.6 0.4 0.0
  0.0 0.0 0.0 0.7 0.3
  0.0 0.0 0.0 0.0 0.0
<EndHMM>
```

- There are two different versions for acoustic model training which depend on whether the sub-word-level (e.g. the phone-level) boundary information exists in the transcription files or not
 - If the training speech files are equipped the sub-word boundaries, i.e., the location of the sub-word boundaries have been marked, the tools *HINIT* and *HREST* can be used to train/generate each sub-word HMM model individually with all the speech training data

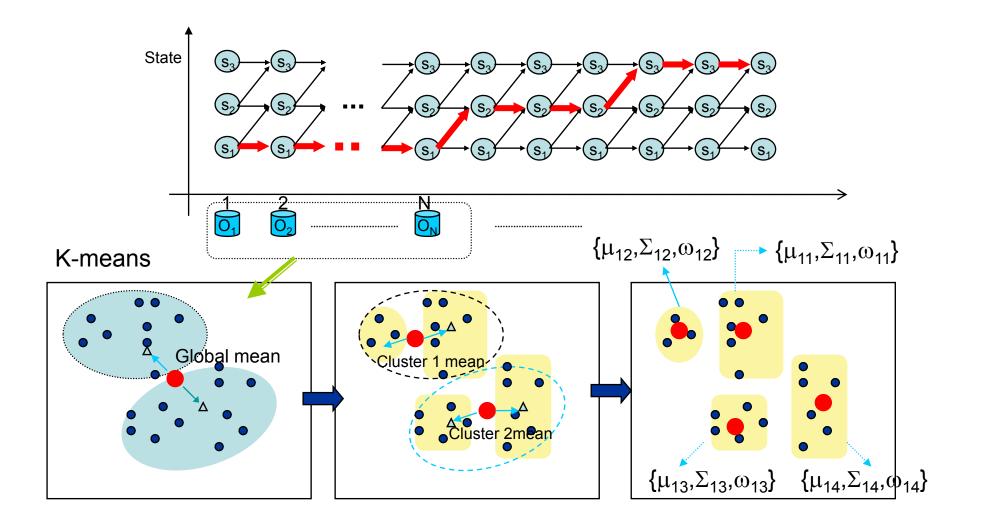


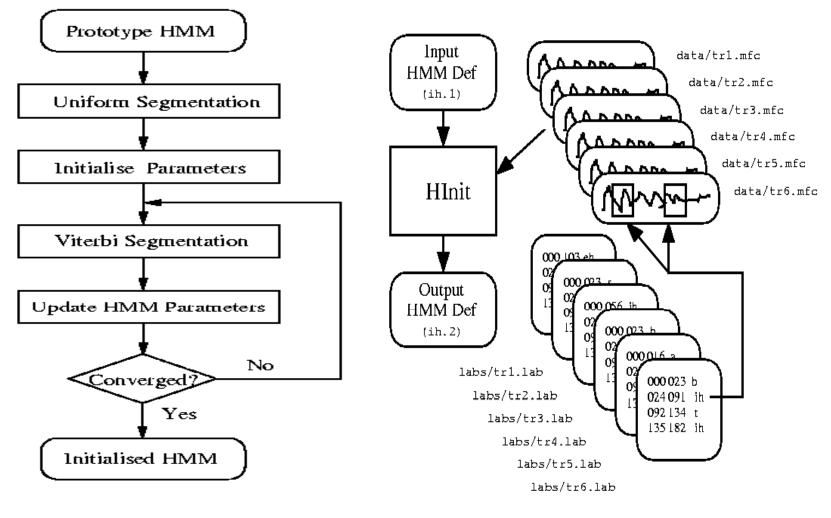
• HINIT

- Iteratively computes an initial set of parameter value using the segmental k-means training procedure
 - It reads in all of the bootstrap training data and cuts out all of the examples of a specific phone
 - On the first iteration cycle, the training data are uniformly segmented with respective to its model state sequence, and each model state matching with the corresponding data segments and then means and variances are estimated. If mixture Gaussian models are being trained, then a modified form of k-means clustering is used
 - On the second and successive iteration cycles, the uniform segmentation is replaced by Viterbi alignment

• HREST

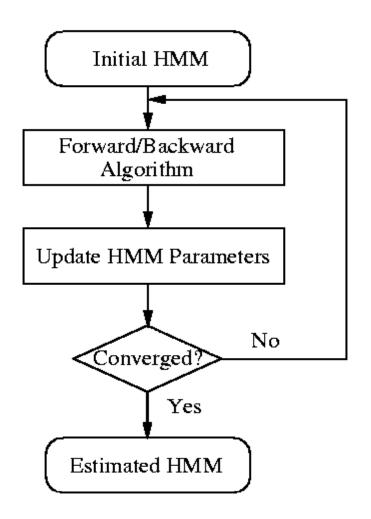
- Used to further re-estimate the HMM parameters initially computed by *HINIT*
- Baum-Welch re-estimation procedure is used, instead of the segmental k-means training procedure for HINIT





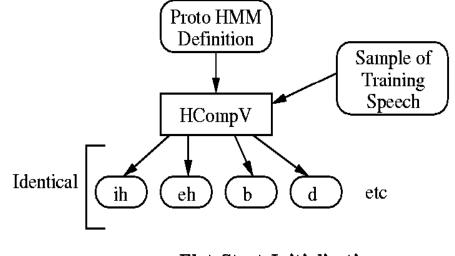
HInit Operation

File Processing in HInit



HRest Operation

- On the other hand, if the training speech files are not equipped the sub-word-level boundary information, a socalled *flat-start* training scheme can be used
 - In this case all of the phone models are initialized to be identical and have state means and variances equal to the global speech mean and variance. The tool HCOMPV can be used for this
- HCOMPV
 - Used to calculate the global mean and variance of a set of training data

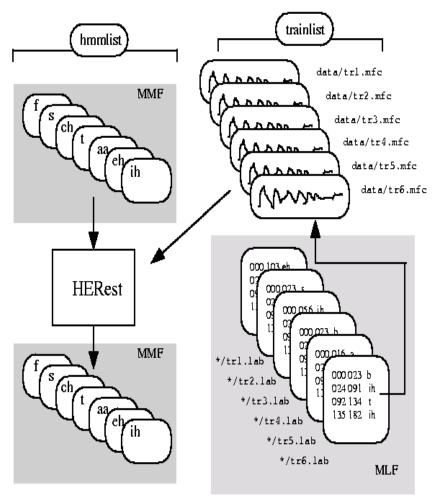


Flat Start Initialisation

 Once the initial parameter set of HMMs has been created by either one of the two versions mentioned above, the tool *HEREST* is further used to perform *embedded training* on the whole set of the HMMs simultaneously using the entire training set

• HEREST

- Performs a single *Baum-Welch* reestimation of the whole set of the HMMs simultaneously
 - For each training utterance, the corresponding phone models are concatenated and the forwardbackward algorithm is used to accumulate the statistics of state occupation, means, variances, etc. for each HMM in the sequence
 - When all of the training utterances has been processed, the accumulated statistics are used to re-estimate the HMM parameters
- HEREST is the core HTK training tool



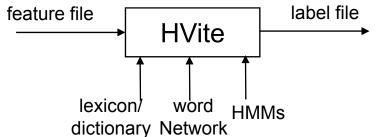
File Processing in HERest

- Model Refinement
 - The philosophy of system construction in HTK is that HMMs should be refined incrementally
 - **CI to CD:** A typical progression is to start with a simple set of single Gaussian context-independent phone models and then iteratively refine them by expanding them to include contextdependency and use multiple mixture component Gaussian $(j_a) \not\leq (au)$ $(j_e) \not\geq (e)$ right-context-dependent modeling distributions

- **Tying:** The tool *HHED* is a HMM definition editor which will clone models into context-dependent sets, apply a variety of parameter tyings and increase the number of mixture components in specified distributions
- Adaptation: To improve performance for specific speakers the tools *HEADAPT* and *HVITE* can be used to adapt HMMs to better model the characteristics of particular speakers using a small amount of training or adaptation data

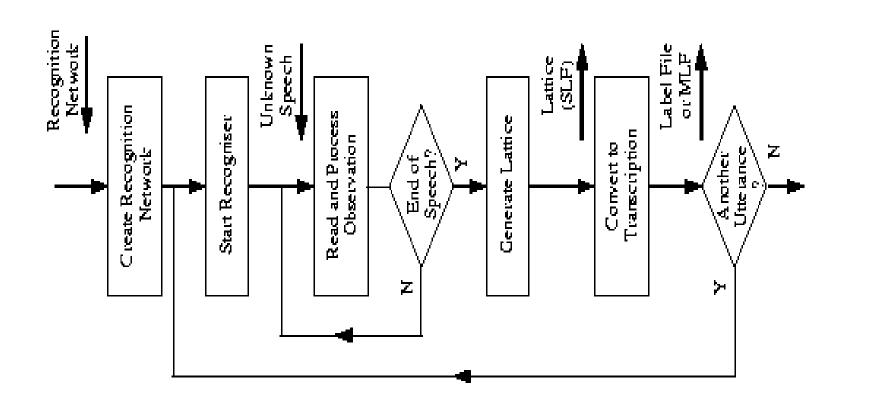
Recognition Phase





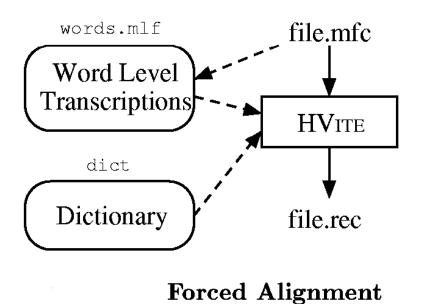
- Performs Viterbi-based speech recognition
- Takes a network describing the allowable word sequences, a dictionary defining how each word is pronounced and a set of HMMs as inputs
- Supports cross-word triphones, also can run with multiple tokens to generate lattices containing multiple hypotheses
- Also can be configured to rescore lattices and perform forced alignments
- The word networks needed to drive HVITE are usually either simple word loops in which any word can follow any other word or they are directed graphs representing a finite-state task grammar
 - *HBUILD* and *HPARSE* are supplied to create the word networks

Recognition Phase (cont.)



Recognition Phase (cont.)

- Generating Forced Alignment
 - HVite computes a new network for each input utterance using the word level transcriptions and a dictionary
 - By default the output transcription will just contain the words and their boundaries. One of the main uses of forced alignment, however, is to determine the actual pronunciations used in the utterances used to train the HMM system



Analysis Phase

- The final stage of the HTK Toolkit is the analysis stage
 - When the HMM-based recognizer has been built, it is necessary to evaluate its performance by comparing the recognition results with the correct reference transcriptions. An analysis tool called *HRESULTS* is used for this purpose
- HRESULTS
 - Performs the comparison of recognition results and correct reference transcriptions by using dynamic programming to align them
 - The assessment criteria of *HRESULTS* are compatible with those used by the US *National Institute of Standards and Technology* (*NIST*)

SP - Berlin Chen 24

test

A Tutorial Example

• A Voice-operated interface for phone dialing

Dial three three two six five four Dial nine zero four one oh nine Phone Woodland Call Steve Young

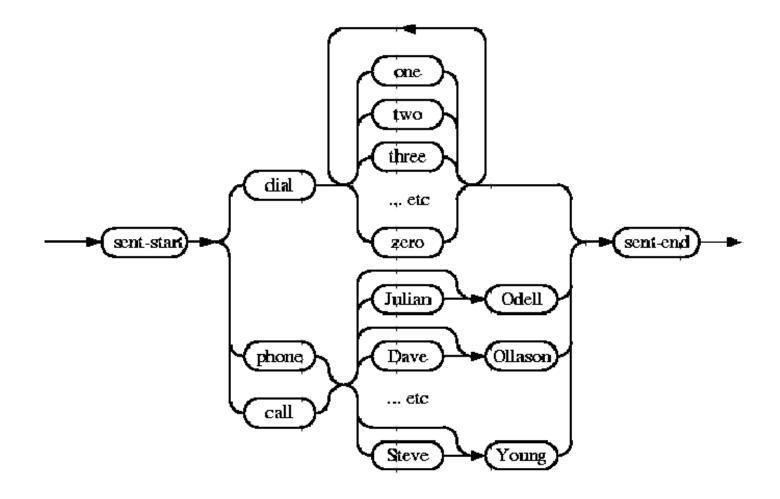
regular expression

\$digit = ONE | TWO | THREE | FOUR | FIVE |
SIX | SEVEN | EIGHT | NINE | OH | ZERO;
\$name = [JOOP] JANSEN | [JULIAN] ODELL | [DAVE]
OLLASON | [PHIL] WOODLAND | [STEVE] YOUNG;

(SENT-START (DIAL <\$digit> | (PHONE|CALL) \$name) SENT-END)

Grammar for Voice Dialing

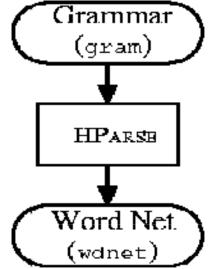
• Grammar for Phone Dialing



Network

- The above high level representation of a task grammar is provided for user convenience
- The HTK recognizer actually requires a word network to be defined using a low level notation called *HTK* Standard Lattice Format (SLF) in which each word instance and each word-to-word transition is listed explicitly

HParse gram wdnet



Dictionary

• A dictionary with a few entries

Α		ah sp
A		ax sp
Α		ey sp
CALL		k ao 1 sp
DIAL		d ay ax l sp
EIGHT		ey t sp
PHONE		f ow n sp
SENT-END	[]	sil
SENT-START	[]	sil
SEVEN		s eh v n sp
то		t ax sp
то		t uw sp
ZERO		z ia r ow sp

- Function words such as A and TO have multiple pronunciations
 The entries
- For SENTSTART and SENTEND have a silence model sil as their pronunciations and null output symbols

Transcription

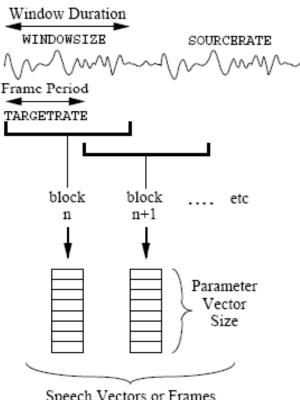
- To train a set of HMMs, every file of training data must have an associated phone level transcription
- Master Label File (MLF)

#!MLF!# "*/S0001.lab" ONE VALIDATED ACTS OF SCHOOL DISTRICTS "*/S0002.lab" TWO OTHER CASES ALSO WERE UNDER ADVISEMENT "*/S0003.lab" BOTH FIGURES (etc.)

Coding The Data



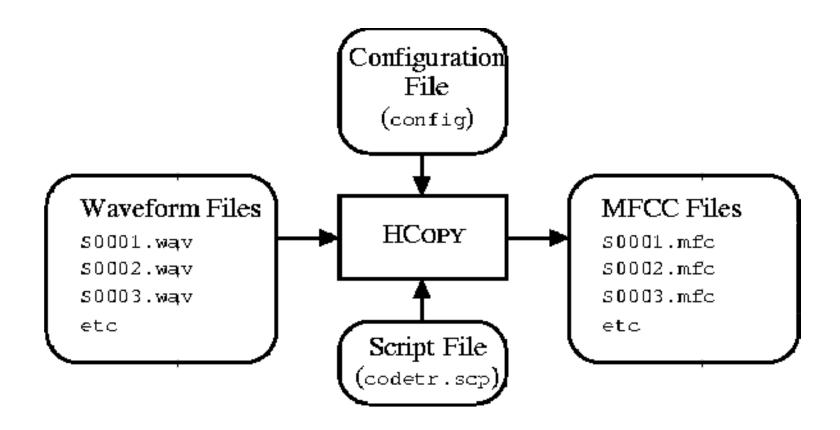
```
in 100 nanosecond unit
# Coding parameters
TARGETKIND =
             MFCC 0
TARGETRATE = 100000.0
                         10ms
SAVECOMPRESSED = T
SAVEWITHCRC = T
                         25ms
WINDOWSIZE = 250000.0
USEHAMMING = T
PREEMCOEF = 0.97
NUMCHANS = 26
CEPLIFTER = 22
NUMCEPS = 12
ENORMALISE = F
```



Speech Vectors or Frames

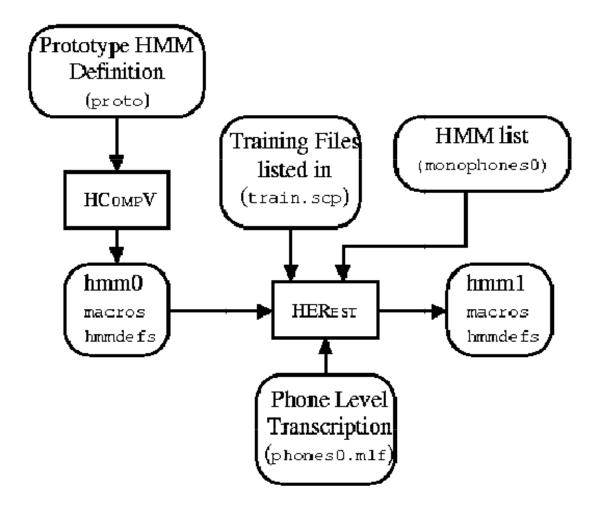
Pre-emphasis filter coefficient Filter bank numbers **Cepstral Liftering Setting** Number of output cepstral coefficients

Coding The Data (cont.)

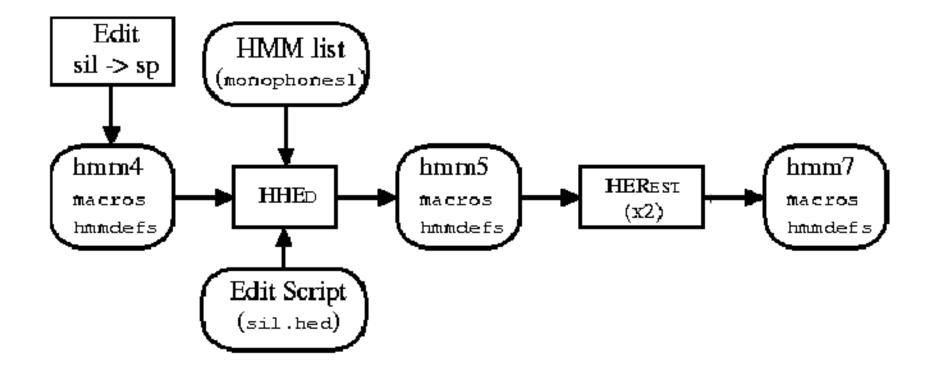


HCopy -T 1 -C config -S codetr.scp

Training



Tee Model



Recognition

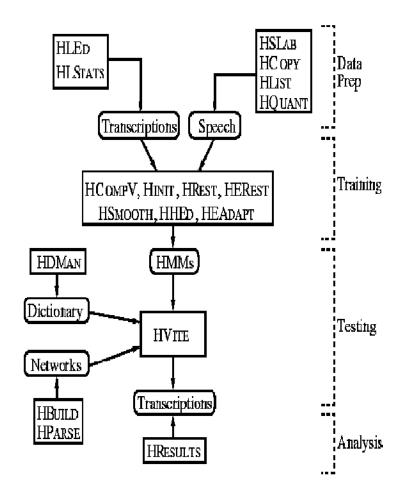
- HVite -T 1 -S test.scp -H hmmset -i results -w wdnet dict hmmlist
- HResults -I refs wlist results

Percent Correct
$$\equiv \frac{N - D - S}{N} \times 100\%$$

Percent Accuracy $\equiv \frac{N - D - S - I}{N} \times 100\%$

Exercises on HTK

- Practice the use of HTK
- Five Major Steps
 - Environment Setup
 - Data Preparation
 HCopy
 - Training
 HHed, HCompV, HErest
 Or Hinit, HHed, HRest, HERest
 - Testing/Recognition
 HVite
 - Analysis
 HResults



Experimental Environment Setup

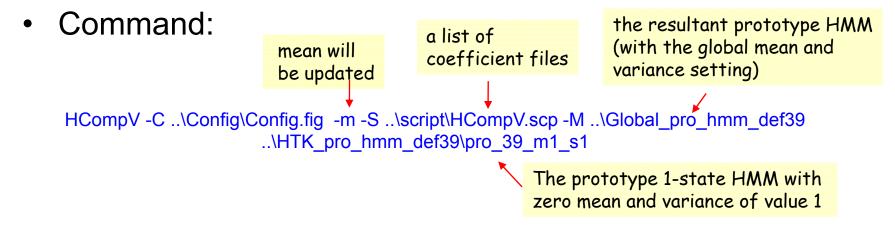
- Download the HTK toolkit and install it
- Copy zipped file of this exercise to a directory name "HTK_Tutorial", and unzipped the file
- Ensure the following subdirectories have been established (If not, make the subdirectories !)
 - 🔁 Batch 🕴 🗅 Chinese <mark>h cocf_HTK_MF</mark>CC hcoef HTK MFCC test 🛅 Config 🛅 Global pro hmm def39 🛅 HTK_pro_hmm_def39 🗀 Init 🛅 Init pro hmm 🛅 Init_pro_hmm_mixture 🛅 label 🛅 pem 🛅 pom_test | 🛅 pro hmm def39 h Rest 🛅 Rest E 🛅 Rest E1 🛅 Rest E2 🛅 Script 🔁 Syllable 🛅 Syllable Test HTK

Step01_HCopy_Train.bat

- Function:
 - Generate MFCC feature files for the training speech utterances
- Command HCOPY -T 00001 -C ..\config\HCOPY.fig -S ..\script\HCopy Train.scp specify the pcm and coefficient files specify the detailed Level of trace information configuration for and their respective directories feature extraction ./pcm/agl-n3-001.pcm ../coef HTK MFCC/agl-n3-001.cof ./pcm/ag1-n3-002.pcm ../coef HTK MFCC/ag1-n3-002.cof #Coding parameters 2 bytes per user defined wave format ../coef HTK MFCC/ag1-n3-003.cof ./pcm/ag1-n3-003.pcm SOURCEFORMAT=ALEN sample ./pcm/ag1-n3-004.pcm ../coef HTK MFCC/ag1-n3-004.cof file header (set to 0 here) HEADERSIZE=0 1e7/16000 ./pcm/ag1-n3-005.pcm ../coef HTK MFCC/ag1-n3-005.cof in accordance with sampling rate SOURCERATE=625 ./pcm/agl-n3-006.pcm ../coef HTK MFCC/ag1-n3-006.cof TARGETKIND=MFCC Z E D A Z(zero mean), E(Energy), D(delta) ./pcm/ag1-n3-007.pcm ../coef HTK MFCC/agl-n3-007.cof TARGETRATE=100000.0 10e-3 *1e7 A(Delta Delta) ./pcm/ag1-n3-008.pcm ../coef HTK MFCC/agl-n3-008.cof #frameshift 10ms ./pcm/ag1-n3-009.pcm ../coef HTK MFCC/ag1-n3-009.cof SAVECOMPRESSED=F ../coef HTK MFCC/agl-n3-010.cof ./pcm/ag1-n3-010.pcm SAVEWITHCRC=F ../coef HTK MFCC/agl-n3-011.cof ./pcm/agl-n3-011.pcm WINDOWSIZE=320000.0 ./pcm/aql-n3-012.pcm ../coef HTK MFCC/agl-n3-012.cof # framesize = 32ms 32e-3 *1e7 ./pcm/agl-n3-013.pcm ../coef HTK MFCC/agl-n3-013.cof Hamming window USEHAMMING=T ./pcm/agl-n3-014.pcm ../coef HTK MFCC/agl-n3-014.cof Pre-emphasis PREEMCOEF=0.97 ../coef HTK MFCC/agl-n3-015.cof ./pcm/ag1-n3-015.pcm filter bank no NUMCHANS=26 ./pcm/agl-n3-016.pcm ../coef HTK MFCC/agl-n3-016.cof liftering setting CEPLIFTER=22 ./pcm/agl-n3-017.pcm ../coef HTK MFCC/agl-n3-017.cof Cepstral coefficient no -> NUMCEPS=12 ../coef HTK MFCC/agl-n3-018.cof ./pcm/agl-n3-018.pcm ENORMALIZE=T ../pcm/agl-n3-019.pcm ../coef HTK MFCC/agl-n3-019.cof NATURALREADORDER=TRUE ./pcm/ag1-n3-020.pcm ../coef HTK MFCC/ag1-n3-020.cof Intel PC byte Order NATURALWRITEORDER=TRUE

Step02_HCompv_S1.bat

- Function:
 - Calculate the global mean and variance of the training data
 - Also set the prototype HMM



• Similar for the batch instructions

Step02_HCompv_S2.bat Step02_HCompv_S3.bat Step02_HCompv_S4.bat

Generate prototype HMMs with different state numbers

Step02_HCompv_S1.bat (count.)

 Note! You should manually edit the resultant prototype HMMs in the directory "Global_pro_hmm_def39" to remove the row

~h "prot_39_m1_sX"

 Remove the name tags, because these proto HMMs will be used as the prototypes for all the INITIALs, FINALs, and silence models

<pre>~0 <<streaminfo> 1 39 <<tr>cvECSIZE> 39<nulld><mfcc a="" d="" e="" z="">~h "pro 39 m1 s1"<beginhmm><numstates> 3<state> 2<mean> 391.839634e-008 1.182030e-008 -2.191493e-009 -3.174569e-009 -2.986750e-009 2.4<variance> 395.612354e+001 5.309968e+001 6.036213e+001 4.772774e+001 5.011119e+001 4.7409<gconst> 9.671951e+001<transp> 30.000000e+000 1.000000e+000 0.000000e+0000.000000e+000 7.000000e+000 0.000000e+000<endhmm></endhmm></transp></gconst></variance></mean></state></numstates></beginhmm></mfcc></nulld></tr></streaminfo></pre>		~0 <streaminfo> 1 39 <vecsize> 39<nulld><mfcc_e_d_a_z> <beginhmm> <numstates> 3 <state> 2 <mean> 39 1.839634e-008 1.182030e-008 -2.191493e-009 -3.174569e-009 -2.986750e-009 2 <variance> 39 5.612354e+001 5.309968e+001 6.036213e+001 4.772774e+001 5.011119e+001 4.74 <gconst> 9.671951e+001 <transp> 3 0.000000e+000 1.000000e+000 0.000000e+000 0.000000e+000 7.000000e-001 3.000000e-001 0.000000e+000 0.000000e+000</transp></gconst></variance></mean></state></numstates></beginhmm></mfcc_e_d_a_z></nulld></vecsize></streaminfo>

Step03_CopyProHMM.bat

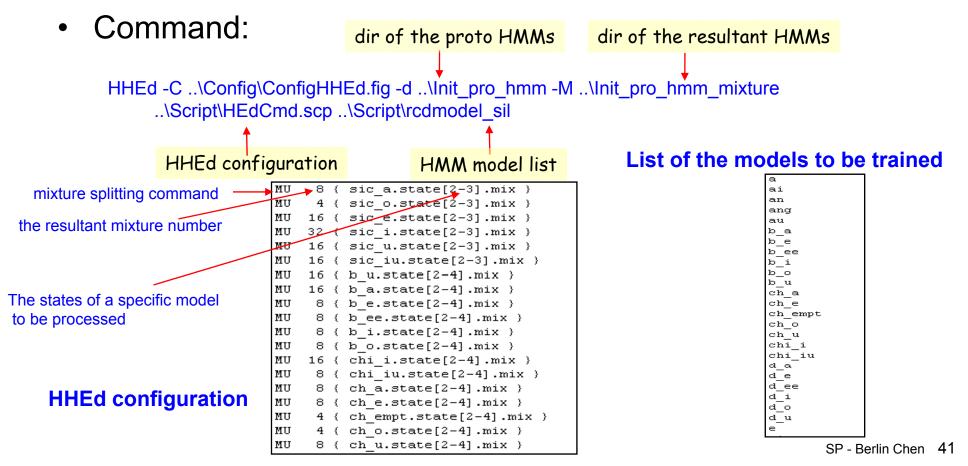
- Function
 - Copy the prototype HMMs, which have global mean and variances setting, to the corresponding acoustic models as the prototype HMMs for the subsequent training process

• Content of the bath file

copy	\Global_pro_hmm_def39\pro_39_m1_s2	\Init_pro_hmm\sic_e
copy	\Global_pro_hmm_def39\pro_39_m1_s2	\Init_pro_hmm\sic_a
copy	\Global_pro_hmm_def39\pro_39_m1_s2	\Init_pro_hmm\sic_o
copy	\Global_pro_hmm_def39\pro_39_m1_s2	\Init_pro_hmm\sic_i
copy	\Global_pro_hmm_def39\pro_39_m1_s2	\Init_pro_hmm\sic_u
copy	\Global_pro_hmm_def39\pro_39_m1_s2	\Init_pro_hmm\sic_iu
copy	\Global_pro_hmm_def39\pro_39_m1_s4	\Init_pro_hmm\u
copy	\Global_pro_hmm_def39\pro_39_m1_s4	\Init_pro_hmm\ua
copy	\Global_pro_hmm_def39\pro_39_m1_s4	\Init_pro_hmm\uai
copy	\Global_pro_hmm_def39\pro_39_m1_s4	\Init_pro_hmm\uan
copy	\Global_pro_hmm_def39\pro_39_m1_s4	\Init_pro_hmm\uang
copy	\Global_pro_hmm_def39\pro_39_m1_s4	\Init_pro_hmm\uei
copy	\Global_pro_hmm_def39\pro_39_m1_s4	\Init_pro_hmm\uen
copy		\Init_pro_hmm\ueng
copy		\Init_pro_hmm\uo
copy	\Global_pro_hmm_def39\pro_39_m1_s4	\Init_pro_hmm\a
copy	\Global_pro_hmm_def39\pro_39_m1_s4	\Init_pro_hmm\ai
copy	\Global_pro_hmm_def39\pro_39_m1_s4	\Init_pro_hmm\an
copy		\Init_pro_hmm\ang
copy	\Global_pro_hmm_def39\pro_39_m1_s4	\Init_pro_hmm\au
сору	\Global_pro_hmm_def39\pro_39_m1_s4	\Init_pro_hmm\e
copy	\Global_pro_hmm_def39\pro_39_m1_s4	\Init_pro_hmm\ei
copy	\Global_pro_hmm_def39\pro_39_m1_s4	\Init_pro_hmm\en
copy		\Init_pro_hmm\eng
copy	\Global_pro_hmm_def39\pro_39_m1_s4	\Init_pro_hmm\er

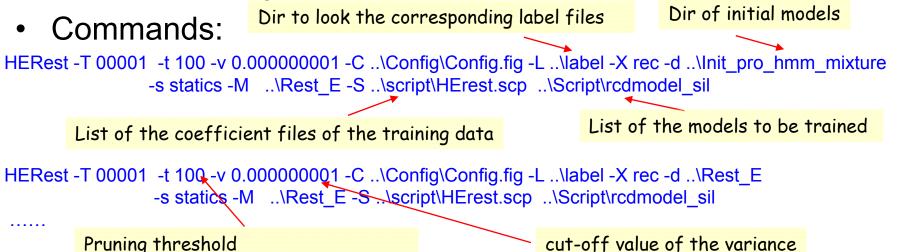
Step04_HHed_ModelMixSplit.bat

- Function:
 - Split the single Gaussian distribution of each HMM state into n mixture of Gaussian distributions, while the mixture number is set with respect to size of the training data for each model



Step05_HERest_Train.bat

- Function:
 - Perform HMM model training
 - Baum-Whelch (EM) training performed over each training utterance using the composite model



of the forward-backward procedures

 You can repeat the above command multiple times, e.g., 30 time, to achieve a better set of HMM models

Step05_HERest_Train.bat (cont.)

A label file of a training utterance

0 1100000 sil 1100000 2800000 b_o 2800000 3600000 o 3600000 4800000 l_u 4800000 6500000 uan 6500000 7300000 f_a 7300000 8800000 en 8800000 10200000 j_e 10200000 11400000 eng 11400000 15900000 sil

Boundary information of the segments of HMM models (will not be used for HERest)

List of the models to be trained

a ai an ang au b a b e b ee b i bо b u ch a ch e ch empt ch o ch u chi i chi iu d a d e d ee d_i d o d u e

Step06_HCopyTest.bat

- Function:
 - Generate MFCC feature files for the testing speech utterances

• Command

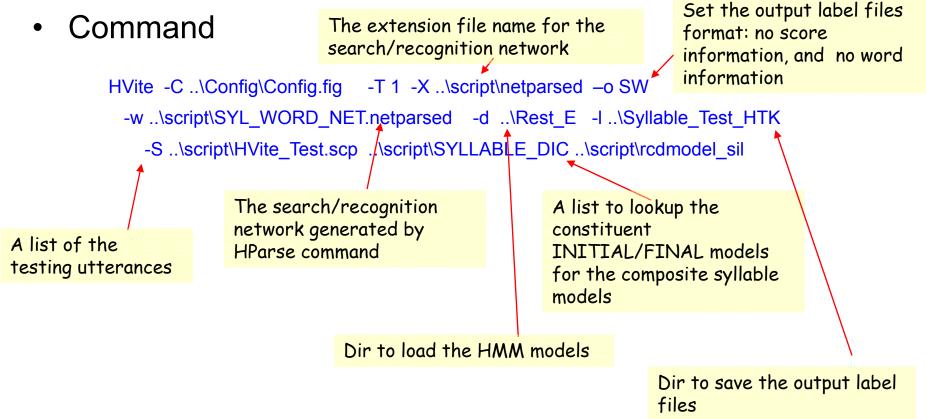
HCOPY -T 00001 -C ..\Config\Config.fig -S ..\script\HCopy_Test.scp

The detailed explanation can be referred to:

Step01_HCopy_Train.bat

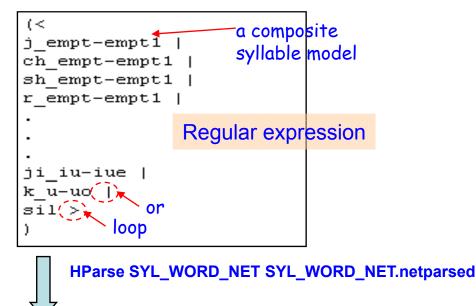
Step07_HVite_Recognition.bat

- Function:
 - Perform free-syllable decoding on the testing utterances



Step07_HVite_Recognition.bat (cont.)

The search/recognition network before performing HParse command



VERSION=1.0 N=407 L=1213 I = OW=!NULL T = 1W=!NULL I=2 W=j empt-empt1 I=3W=!NULL I=4W=ch empt-empt1 I=5W=sh empt-empt1 I = 6W=r empt-empt1 I=7W=tz empt-empt2 I=8 W=ts empt-empt2 I=9 W=s empt-empt2 I = 10W=sic a-a I=11W=j a-a

composite syllable models j_empt-empt1 j_empt empt1

A list to lookup the constituent

INITIAL/FINAL models for the

ch_empt-empt1 ch_empt empt1
•
•
sic_a-a sic_a a
j_a-a j_a a
ch_a-a ch_a a
sh_a-a sh_a a
tz_a-a tz_a a
•
•
sil sil

The search/recognition network generated by HParse command

Step08_HResults_Test.bat

- Function:
 - Analyze the recognition performance
- Command The extension file name for the label files

ignore the silence label "sil"

HResults -C ..\Config\Config.fig -T 00020 -X rec -e ??? sil -L ..\Syllable

-S ..\script\Hresults_rec600.scp ..\script\SYLLABLE_DIC

A list of the label files generated by the recognition process

Dir lookup the reference label files

Syllable Test HTK\bruce-b07-001.rec Syllable Test HTK\bruce-b07-002.rec Syllable Test HTK\bruce-b07-003.rec Syllable Test HTK\bruce-b07-004.rec Syllable Test HTK\bruce-b07-005.rec Syllable Test HTK\bruce-b07-006.rec Syllable Test HTK\bruce-b07-007.rec Syllable Test HTK\bruce-b07-008.rec Syllable Test HTK\bruce-b07-009.rec Syllable Test HTK\bruce-b07-010.rec Syllable Test HTK\bruce-b07-011.rec Syllable Test HTK\bruce-b07-012.rec Syllable Test HTK\bruce-b07-013.rec Syllable Test HTK\bruce-b07-014.rec Syllable Test HTK\bruce-b07-015.rec Syllable Test HTK\bruce-b07-016.rec Syllable Test HTK\bruce-b07-017.rec Syllable Test HTK\bruce-b07-018.rec Syllable Test HTK\bruce-b07-019.rec Syllable Test HTK\bruce-b07-020.rec

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Date: Sat Nov 22 10:02:13 2003		
Ref :/Syllable		
Rec :\Syllable_Test_HTK\bruce-b07-001.rec		
:\Syllable_Test_HTK\bruce-b07-002.rec		
:\Syllable_Test_HTK\bruce-b07-003.rec		
:\Syllable_Test_HTK\bruce-b07-004.rec		
:\Syllable_Test_HTK\bruce-b07-005.rec		
Overall Results		
SENT: %Correct=11.33 [H=68, S=532, N=600]		
WORD: %Corr=69.49, Acc=64.16 [H=2685, D=47, S=1132, I=206, N=3864]		
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Step09_BatchMFCC_Def39.bat

• Also, you can train the HMM models in another way

Hinit \longrightarrow (HHEd) \implies HRest \implies HERest

- For detailed information, please referred to the previous slides or the HTK manual
- You can compare the recognition performance by running

Step02~Step05

or Step09 alone