List of modifications made in HTS (for version 2.3 beta)

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December 25, 2014

1 Modifications in Model Definition

In HTS, the HTK HMM definition (please see HTKBook [1] Chapter 7) has been modified to support MSD [2], stream-level tying, and adaptation of multi-stream HMMs. This section gives its brief description.

First, <MSDInfo> is added to global options of the HTK HMM definition language The arguments to the <MSDInfo> option are the number of streams (default 1) and then for each stream, 0 (non-MSD stream) or 1 (MSD stream) of that stream. The full set of global options in HTS is given below.

```
globalOpts  =  option { option }
option      =  <HmmSetId> string |
               <StreamInfo> short { short } |
               <MSDInfo> short { short } |
               <VecSize> short |
               <ProjSize> short |
               <InputXform> inputXform |
               <ParentXform> ~a macro |
               covkind |
               durkind |
               parmkind
```

Second, the number of mixture specification is modified to support stream-level tying structure as follows:

```
HTK                             HTS
 <State> 2                       <State> 2
  <NumMixes> 1 2
  <SWeights> 2 0.9 1.1           <SWeights> 2 0.9 1.1
  <Stream> 1                     <Stream> 1
                                    <NumMixes> 1
    <Mixture> 1 1.0               <Mixture> 1 1.0
     <Mean> 4                      <Mean> 4
       0.3 0.2 0.1 0.0              0.3 0.2 0.1 0.0
     <Variance> 4                  <Variance> 4
       0.5 0.4 0.3 0.2              0.5 0.4 0.3 0.2
   <Stream> 2                    <Stream> 2
                                   <NumMixes> 2
   <Mixture> 1 0.4               <Mixture> 1 0.4
    <Mean> 2                      <Mean> 2
      1.0 2.0                      1.0 2.0
    <Variance> 2                  <Variance> 2
      4.0 8.0                      4.0 8.0
   <Mixture> 2 0.6               <Mixture> 2 0.6
    <Mean> 2                      <Mean> 2
      2.0 9.0                      2.0 9.0
    <Variance> 2                  <Variance> 2
      3.0 6.0                       3.0 6.0
```

As you can see, <NumMixes> is moved from state-level to stream-level. This modification enables us to include the number of mixture component in the stream-level macro. Based on this implementation, stream-level macro was added. The various distinct points in the hierarchy of HMM parameters which can be tied in HTS is as follows:

~s	shared state distribution
~p	shared stream
~m	shared Gaussian mixture component
~u	shared mean vector
~v	shared diagonal variance vector
~i	shared inverse full covariance matrix
~c	shared Cholesky U matrix
~x	shared arbitrary transform matrix
~t	shared transition matrix
~d	shared duration parameters
~w	shared stream weight vector

Note that the ~p macro is used by the HMM editor HHED for building tied mixture systems in the original HTK macro definition.

The resultant state definition of in the modified HTK HMM definition language is as follows:

state	=	<State> short stateinfo
stateinfo	=	~s macro |
		[weights] stream { stream } [duration]
macro	=	string
weights	=	~w macro | <SWeights> short vector
vector	=	float { float }
stream	=	[<Stream> short] streaminfo
streaminfo	=	~p macro | [<Stream> short] [mixes] (mixture { mixture } | tmixpdf | discpdf)
mixes	=	<NumMixes> short {short}
tmixpdf	=	<TMix> macro weightList
weightList	=	repShort { repShort }
repShort	=	short [∗ char]
discpdf	=	<DProb> weightList
mixture	=	[<Mixture> short float] mixpdf
mixpdf	=	~m macro | mean cov [<GConst> float]
mean	=	~u macro | <Mean> short vector
cov	=	var | inv | xform
var	=	~v macro | <Variance> short vector
inv	=	~i macro |
		(<InvCovar> | <LLTCovar>) short tmatrix
xform	=	~x macro | <Xform> short short matrix
matrix	=	float {float}
tmatrix	=	matrix

It should be noted that <Stream> can doubly be specified in both stream and streaminfo. This is because <Stream> in ~p macro is essential to specify stream index of this macro. This stream index information is used in various HTS functions to check stream consistency.

Third, to support multi-stream HMM adaptation, the HTK HMM definition language for baseclasses is modified. A baseclass is defined as

baseClass	=	~b macro baseopts classes
baseopts	=	<MMFIdMask> string <Parameters> baseKind [<StreamInfo>] <NumClasses> int
StreamInfo	=	short { short } |
baseKind	=	MIXBASE | MEANBASE | COVBASE
classes	=	<Class> int itemlist { classes }

where <StreamInfo> is optionally added to specify the stream structure.

2 Added Configuration Variables

A number of configuration variables have been added to HTK to control new functions implemented in HTS. Their names, default values, and brief descriptions are as follows:

Module	Name	Default	Description
HADAPT	SAVEFULLC	F	Save transformed model set in full covariance form
	USEMAPLR	F	Use MAP criterion [3]
	USEVBLR	F	Use VB criterion [4]
	USESTRUCTURALPRIOR	F	Perform structural approach (e.g., SMAPLR, CSMAPLR, and SVBLR)
	PRIORSCALE	1.0	Prior parameter for MAPLR and VBLR
	SAVEALLNODEXFORMS	T	Save all (unnecessary) linear transforms estimated in SMAPLR/CSMAPLR/SVBLR
	BANDWIDTH		Bandwidth of transformation matrices [5]
	DURUSEBIAS	F	Specify a bias with linear transforms
	DURSPLITTHRESH	1000.0	Minimum occupancy to generate a transform for state duration model set
	DURTRANSKIND	MLLRMEAN	Transformation kind
	DURBLOCKSIZE	full	Block structure of transform for state duration model set
	DURBANDWIDTH		Bandwidth of transformation matrices for state duration model set
	DURBASECLASS	global	Macroname of baseclass for state duration model set
	DURREGTREE		Macroname of regression tree for state duration model set
	DURADAPTKIND	BASE	Use regression tree or base classes to adapt state duration model set
HFB	MAXSTDDEVCOEF	10	Maximum duration to be evaluated
	MINDUR	5	Minimum duration to be evaluated
HMAP	APPLYVFLOOR	T	Apply variance floor to model set
HGEN	MAXEMITER	20	Maximum number of EM iterations
	EMEPSILON	1.0E-4	Convergence factor for EM iteration

Module	Name	Default	Description
	RNDPARMEAN	0.0	Mean of Gaussian noise for random generation [6]
	RNDPARVAR	1.0	Variance of Gaussian noise for random generation
	USEGV	F	Use speech parameter generation algorithm considering GV [7]
	USEGVPST		GV calculation flag for each stream
	CDGV	F	Use context-dependent GV model set
	LOGGV	F	Use logarithmic GV instead of linear GV
	MAXGVITER	F	Max iterations in the speech parameter generation considering GV
	GVEPSILON	1.0E-4	Convergence factor for GV iteration
	MINEUCNORM	1.0E-2	Minimum Euclid norm of a gradient vector
	STEPINIT	1.0	Initial step size
	STEPDEC	0.5	Step size deceleration factor
	STEPINC	1.2	Step size acceleration factor
	HMMWEIGHT	1.0	Weight for HMM output prob
	GVWEIGHT	1.0	Weight for GV output prob
	GVINITWEIGHT	1.0	Initial weight of GV
	OPTKIND	NEWTON	Optimization method
	RNDFLAGS		Random generation flag
	GVMODELMMF		GV MMF file
	GVHMMLIST		GV model list
	GVMODELDIR		Dir containing GV models
	GVMODELEXT		Ext to be used with above Dir
	GVOFFMODEL		Model names to be excluded from GV calculation
	USEDAEM	F	Use the DAEM-based parameter generation algorithm
	DAEMITER	20	Number of iterations for the DAEM-based parameter generation algorithm
	DAEMTEMPSCHEDULE	1.0	Temperature schedule parameter for DAEM
HMODEL	IGNOREVALUE	-1.0E+10	Ignore value to indicate zero-dimensional space in multi-space probability distribution
HCOMPV	NSHOWELEM	12	Number of vector elements to be shows
	VFLOORSCALE	0.0	variance flooring scale
	VFLOORSCALESTR		variance flooring scale vector for streams

Module	Name	Default	Description
HERᴇꜱᴛ	APPLYVFLOOR	T	Apply variance floor to model set
	DURMINVAR	0.0	Minimum variance floor for state duration model set
	DURVARFLOORPERCENTILE	0	Maximum number of Gaussian components (as the percentage of the total Gaussian components in the system) to undergo variance floor for state duration model set
	APPLYDURVARFLOOR	T	Apply variance floor to state duration model set
	DURMAPTAU	0.0	MAP tau for state duration model set [8]
	ALIGNDURMMF		State duration MMF file for alignment (2-model reest)
	ALIGNDURLIST		State duration model list for alignment (2-model reest)
	ALIGNDURDIR		Dir containing state duration models for alignment (2-model reest)
	ALIGNDUREXT		Ext to be used with above Dir (2-model reest)
	ALIGNDURXFORMEXT		Input transform ext for state duration model set to be used with 2-model reest
	ALIGNDURXFORMDIR		Input transform dir for state duration model set to be used with 2-model reest
	DURINXFORMMASK		Input transform mask for state duration model set (default output transform mask)
	DURPAXFORMMASK		Parent transform mask for state duration model set (default output parent mask)
HHEᴅ	USEPATTERN	F	Use pattern instead of base phone for tree-based clustering
	SINGLETREE	F	Construct single tree for each state position
	APPLYMDL	F	Use the MDL criterion for tree-based clustering [9]
	IGNORESTRW	F	Ignore stream weight in tree-based clustering
	REDUCEMEM	F	Use reduced memory implementation of tree-based clustering
	MINVAR	1.0E-6	Minimum variance floor for model set

Module	Name	Default	Description
	MDLFACTOR	1.0	Factor to control the model complexity term in the MDL criterion
	MINLEAFOCC	0.0	Minimum occupancy count in each leaf node
	MINMIXOCC	0.0	Minimum occupancy count in each mixture component
	SHRINKOCCTHRESH		Minimum occupancy count in decision trees shrinking
HMGENS	SAVEBINARY	F	Save generated parameters in binary
	OUTPDF	F	Output pdf sequences
	PARMGENTYPE	0	Type of parameter generation algorithm [10]
	MODELALIGN	F	Use model-level alignments given from label files to determine model-level durations
	STATEALIGN	F	Use state-level alignments given from label files to determine state-level durations
	USEALIGN	F	Use model-level alignments to prune EM-based parameter generation algorithm
	USEHMMFB	F	Do not use state duration models in the EM-based parameter generation algorithm
	INXFORMMASK		Input transform mask
	PAXFORMMASK		Parent transform mask
	PDFSTRSIZE		Number of PdfStreams
	PDFSTRORDER		Size of static feature in each PdfStream
	PDFSTREXT		Ext to be used for generated parameters from each PdfStream
	WINEXT		Ext to be used for window coefficients file
	WINDIR		Dir containing window coefficient files
	WINFN		Name of window coefficient files
HSMMALIGN	INXFORMMASK		Input transform mask
	PAXFORMMASK		Parent transform mask
	DURINXFORMMASK		Input transform mask for state duration model set (default output transform mask)
	DURPAXFORMMASK		Parent transform mask for state duration model set (default output parent mask)

Module	Name	Default	Description
HMGETOOL	SAVEBINARY	F	Save estimated models in binary
	PDFSTRSIZE		Number of PdfStreams
	PDFSTRORDER		Size of static feature in each Pdf-Stream
	MGETRNFLAG		Whether perform MGE train for the stream
	GVTRNFLAG		Whether incorporate GV component into MGE training
	ACCERRFLAG		Accumulate generation error
	INVQUASIZE		Bandwith of quasi-diagonal inversion matrix
	VARWINSIZE		Window for local variance calculation
	GVDISTWGHT		GV weights
	GAINWTFLAG		gain weight for generation error
	WINEXT		Ext to be used for window coefficients file
	WINDIR		Dir containing window coefficient files
	WINFN		Name of window coefficient files

Other configuration variables in HTK can also be used with HTS. Please refer to HTKBook [1] Chapter 18 for others.

3 Added Command-Line Options

Various new command-line options have also been added to HTK tools. They are listed as follows:

HINIT

```
Option                                                Default
 -g      Ignore outlier vector in MSD                   on
```

HREST

```
Option                                                Default
 -g s    output duration model to file s                none
 -o fn   Store new hmm def in fn (name only)            outDir/srcfn
```

HEREST

```
Option                                                Default
 -b      use an input linear transform for dur models   off
 -f s    extension for new duration model files         as src
 -g s    output duration model to file s                none
 -k f    set temperature parameter for DAEM training    1.0
 -n s    dir to find duration model definitions         current
 -q s    save all xforms for duration to TMF file s     TMF
 -u tmvwapd update t)rans m)eans v)ars w)ghts            tmvw
                   a)daptation xform p)rior used
                   s)emi-tied xform
                   d) switch to duration model update flag
 -y s    extension for duration model files             none
 -N mmf  load duration macro file mmf
 -R dir  dir to write duration macro files              current
 -W s [s] set dir for duration parent xform to s        off
         and optional extension
 -Y s [s] set dir for duration input xform to s         none
         and optional extension
 -Z s [s] set dir for duration output xform to s        none
```

HHED

```
Option                                                Default
 -a f    factor to control the second term in the MDL   1.0
 -i      ignore stream weight                           off
 -m      apply MDL principle for clustering             off
 -p      use pattern instead of base phone              off
 -q n    use reference tree for clustering              off
           0: clustering is stopped by threshold
           1: clustering is stopped when leaf don't have occ
           2: clustering is stopped by threshold
              After that, standard clustering is not performed
           3: clustering is stopped when leaf don't have occ
```

```
               After that, standard clustering is not performed
  -r n    reduce memory usage on clustering                   0
            0: no memory reduction
            1: mid reduction but fast
            2: large reduction but slow
  -s      construct single tree                              off
  -v f    Set minimum variance to f                          1.0E-6
```

HMGENS

```
Option                                                    Default
-a      Use an input linear transform for HMMs            off
-b      Use an input linear transform for dur models      off
-c n    type of parameter generation algorithm            0
            0: both mix and state sequences are given
            1: state sequence is given,
               but mix sequence is hidden
            2: both state and mix sequences are hidden
-d s    dir to find hmm definitions                       current
-e      use model alignment from label for pruning        off
-f f    frame shift in 100 ns                              50000
-g f    Mixture pruning threshold                          10.0
-h s [s] set speaker name pattern to s,                   *.%%%
         optionally set parent patterns
-m      use model alignment for duration                  off
-n s    dir to find duration model definitions            current
-p      output pdf sequences                               off
-r f    speaking rate factor (f<1: fast  f>1: slow)       1.0
-s      use state alignment for duration                  off
-t f [i l] set pruning to f [inc limit]                   inf
-v f    threshold for switching spaces for MSD            0.5
-x s    extension for hmm files                            none
-y s    extension for duration model files                none
-A      Print command line arguments                       off
-B      Save HMMs/transforms as binary                     off
-C cf   Set config file to cf                              default
-D      Display configuration variables                    off
-E s [s] set dir for parent xform to s                     off
         and optional extension
-G fmt  Set source label format to fmt                     as config
-H mmf  Load HMM macro file mmf
-I mlf  Load master label file mlf
-J s [s] set dir for input xform to s                      none
         and optional extension
-L dir  Set input label (or net) dir                      current
-M dir  Dir to write HMM macro files                      current
-N mmf  Load duration macro file mmf
-S f    Set script file to f                               none
-T N    Set trace flags to N                               0
-V      Print version information                          off
-W s [s] set dir for duration parent xform to s            off
         and optional extension
```

```
-X ext  Set input label (or net) file ext              lab
-Y s [s] set dir for duration input xform to s         none
        and optional extension
```

HSMMALIGN

```
Option                                                 Default
-a       Use an input linear transform for HMMs        off
-b       Use an input linear transform for dur models  off
-c       Prune by time information of label            off
-d s     Dir to find hmm definitions                   current
-f       Output full state alignment                   off
-h s [s] Set speaker name pattern to s,                *.%%%
        optionally set parent patterns
-n s     Dir to find duration model definitions        current
-m dir   Set output label dir                          current
-r ext   Output label file extension                   lab
-s s     print statistics to file s                    off
-t i     Set pruning threshold                         off
-w f     Duration weight                               1.0
-x s     Extension for hmm files                       none
-y s     Extension for duration model files            none
-A       Print command line arguments                  off
-C cf    Set config file to cf                         default
-D       Display configuration variables               off
-E s [s] set dir for parent xform to s                 off
        and optional extension
-F fmt   Set source data format to fmt                 as config
-G fmt   Set source label format to fmt                as config
-H mmf   Load HMM macro file mmf
-I mlf   Load master label file mlf
-J s [s] set dir for input xform to s                  none
        and optional extension
-L dir   Set input label (or net) dir                  current
-N mmf   Load duration macro file mmf
-S f     Set script file to f                          none
-T N     Set trace flags to N                          0
-V       Print version information                     off
-W s [s] set dir for duration parent xform to s        off
        and optional extension
-X ext   Set input label (or net) file ext             lab
-Y s [s] set dir for duration input xform to s         none
        and optional extension
```

HMGETOOL

```
Option                                                 Default
-a i j  i: max times to shift the boundary             none
        j: max length for each boundary shifting       1
-b i j  i: end iteration for boundary adjustment       none
        j: window size for boundary adjustment         5
-c      output the process data                        off
```

```
-d dir   HMM definition directory                        none
-e       enable limit the updating rate for each step    off
-f r     frame rate                                      50000
-g       enable multiply variance ratio for mean updating off
-i i j   start/end iteration index of MGE training       0 0
-j flg   0: eval 1: train 2: adapt                       1
-l dir   output label directory                          none
-o ext   HMM def file extension                          none
-p a b   parameter for step size: 1/(a + b*n)            1000.0 1.0
-r file load HMM for reference                           none
-s file updating scale file                              none
-u mvwa update t)rans m)eans v)ars w)ghts                none
              a)daptation xform
-v f     threshold for switching spaces for MSD          0.5
-w f     distance weight for gv component                1.0
-x ext   label file extension                            lab
-A       Print command line arguments                    off
-B       Save HMMs/transforms as binary                  off
-C cf    Set config file to cf                           default
-D       Display configuration variables                 off
-G fmt   Set source label format to fmt                  as config
-H mmf   Load HMM macro file mmf
-I mlf   Load master label file mlf
-J s [s] set dir for input xform to s                    none
         and optional extension
-K s [s] set dir for output xform to s                   none
         and optional extension
-L dir   Set input label (or net) dir                    current
-M dir   Dir to write HMM macro files                    current
-S f     Set script file to f                            none
-T N     Set trace flags to N                            0
-V       Print version information                       off
-X ext   Set input label (or net) file ext              lab
```

Please also refer to HTKBook [1] Chapter 17 for other command-line options.

4 Added Commands and Modifications in HHED

Some HHED commands have been added in HTS. They are as follows:

```
AX filename          - Set the Adapt XForm to filename
AX filename state_mapping_table mmf list output_filename
                     - Set the Adapt XForm to another model and save one
CM directory         - Convert models to pdf for speech synthesizer
CT directory         - Convert trees/questions for speech synthesizer
DM type macroname    - Delete macro from model-set
DR id                - Convert decision trees to a regression tree
DV                   - Convert full covariance to diagonal variances
IT filename          - Clustering while imposing loaded tree structure
                        If any empty leaf nodes exist, loaded trees
                        are pruned
                        and then saved to filename
IX filename          - Set the Input Xform to filename
JM hmmFile itemlist  - Join Models on stream or state level
PX filename          - Set the Parent Xform to filename
SM smtable mmf hlist - Output KLD-based state mapping table
// comment           - Comment line (ignored)
```

In many HHED commands, we are required to specify item lists to specify a set of items to be processed. In HTS, item list specification has been modified to specify stream-level items.

```
itemList    = "{" itemSet { "," itemSet } "}"
itemSet     = hmmName . ["transP" | "state" state ]
hmmName     = ident | identList
identList   = "(" ident { "," ident } ")"
ident       = < char | metachar >
metachar    = "?" | "⋆"
state       = index ["." stateComp ]
index       = "[" intRange { "," intRange } "]"
intRange    = integer [ "-" integer ]
stateComp   = "dur" | "weights" | stream
stream      = [ " stream" index ] [ ".mix" mix ]
mix         = index [ "." ( "mean" | "cov" ) ]
```

For example,

```
  TI str1 {*.state[2].stream[1]}
```

denotes tying streams in state 2 of all phonemes.

Appendix A   History of HTS

- **Version 1.0 (December 2002)**
  - Based on HTK-3.2.
  - HHED supports tree-based clustering based on the MDL criterion [9].
  - HHED supports stream-dependent tree-based clustering [11].
  - HMODEL supports multi-space probability distributions (MSD) [2].
  - HEREST can generate state duration modeling [12].
  - Speech parameter generation algorithm [10] is implemented in HGEN and HMGENS.
  - Demo using the CMU Communicator database.
- **Version 1.1 (May 2003)**
  - Based on HTK-3.2.
  - Small run-time synthesis engine (hts_engine).
  - Demo using the CSTR TIMIT database.
  - HTS voices for the Festival speech synthesis system [13].
- **Version 1.1.1 (December 2003)**
  - Based on HTK-3.2.1.
  - HCOMPV supports variance flooring for MSD-HMMs.
  - Demo using the CMU ARCTIC database [14].
  - Demo using the Nitech Japanese database.
  - Demo supports post-filtering [15].
  - HTS voice for the Galatea toolkit [16].
- **Version 2.0 (December 2006) [17]**
  - Based on HTK-3.4.
  - Support generating state duration PDFs in HREST.
  - Phoneme boundaries can be given to HEREST using the -e option [18].
  - Reduced-memory implementation of tree-based clustering in HHED with the -r option.
  - Each decision tree can have a name with regular expressions in HHED with the -p option.
  - Flexible model structures in HMGENS.
  - Speech parameter generation algorithm based on the EM algorithm [10] in HMGenS.
  - Random generation algorithm [6] in HMGENS [6].
  - State or phoneme-level alignments can be given to HMGENS.
  - The interface of HMGENS has been switched to HEREST-style.
  - Various kinds of linear transformations for MSD-HMMs are supported in HADAPT.
    * Constrained MLLR based adaptation [19].
    * Adaptive training based on constrained MLLR [19].
    * Precision matrix modeling based on semi-tied covariance matrices [20].
    * Heteroscedastic linear discriminant analysis (HLDA) based feature transform [21].
    * Phonetic decision trees can be used to define regression classes for adaptation [22]
    * Adapted HMMs can be converted to the run-time synthesis engine format.
  - Maximum a posteriori (MAP) adaptation [8] for MSD-HMMs in HMAP.
- **Version 2.0.1 (May 2007)**
  - Based on HTK-3.4.
  - HADAPT supports band structure for linear transforms [5].
  - HCOMPV supports stream-dependent variance flooring scales.
  - Demo support LSP-type spectral parameters.
  - β version of the runtime synthesis engine API (hts_engine API).
  - hts_engine API supports speaker interpolation [23].
- **Version 2.1 (June 2008)**
  - Based on HTK-3.4.
  - Released under the Modified BSD license [24].

- Simple documentation.
- 64-bit compile support.
- MAXSTRLEN (maximum length of strings), MAXFNAMELEN (maximum length of filenames), PAT_LEN (maximum length of pattern strings), and SMAX (maximum number of streams) defined in HShell.h can be set through configure script.
- HFB supports the forward/backward algorithm for hidden semi-Markov models (HSMMs) [25, 26].
- HADAPT suports SMAPLR/CSMAPLR adaptation [27, 3].
- HGEN supports speech parameter generation algorithm considering global variance (GV) [7].
- HGEN supports random generation of transitions, durations, and mixture components.
- HEREST supports HSMM training and adaptation.
- HMGENS supports speech parameter generation from HSMMs.
- Add DM command to HHED to delete an existing macro from MMF.
- Add IT command to HHED to impose pre-constructed trees in clustering.
- Add JM command to HHED to join models on state or stream level.
- HHED MU command supports '*2' style mixing up.
- HHED MU command supports mixture-level occcupancy threshhold in mixing up.
- First stable version of the runtime synthesis engine API (hts_engine API).
- **Version 2.1.1 (May 2010)**
  - Based on HTK-3.4.1
  - WFST converter for forced-alignment of HSMM is implemented in HFST [28].
  - Demo support context-dependent GV without silent and pause phoneme.
  - Add initial GV weight for parameter generation in HMGENS.
  - Add memory reduction options for context-clustering in HHED.
  - Add model-level alignments given from label of singing voice to determine note-level durations.
  - Demo using the Nitech Japanese database for singing voice synthesis [29].
  - The API of runtime synthesis engine, hts_engine API, is splitted from HTS itself and moved to SourceForge.
- **Version 2.2 (July 2011)**
  - Support DAEM algorithm in parameter estimation step [30].
  - Support KLD-based state-mapping and cross-lingual speaker adaptation in HHED [31].
  - Add stand-alone HSMM-based forced-alignment command, HSMMALIGN, instead of HFST [28].
  - Add HMGETOOL for MGE training [32].
  - Context-clustering can be started in the middle of the tree building.
  - Change sampling frequency of demo from 16kHz to 48kHz.
  - Support bark critical-band based aperiodic measure.
  - Change speakers of Brazilian Portuguese and Japanese song demo.
  - Release slides as a tutorial of HMM-based speech synthesis.
- **Version 2.3 beta (December 2014)**
  - Add VBLR adaptation in HEREST [4].
  - Add DAEM-based parameter generation algorithm in HMGENS.
  - Support DP search to determine state duration when the model alignments are given in HMGENS.
  - Speed up context-clustering by calculating differences between answers to current and previous questions.
  - Add LSP postfilter in demo [33].
  - Support mel-cepstrum based aperiodic measure in STRAIGHT demo.
  - Support new HTS voice format for hts_engine API.
  - Turn off spectrum normalization in STRAIGHT demo.
  - Add modulation spectrum-based postfilter in demo [34].
  - Add untying weights function in HHEd.
  - Support text files instead of utt files for general English database.
  - Integrate normal demo and STRAIGHT demo.

References

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