

# HMM-based Speech Synthesis System (HTS) - Reference of hts\_engine API

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## Reference of hts\_engine API

- [Preprocessor symbols for conditional compilation](#)
- [Structures](#)
  - [Models](#)
    - [HTS\\_ModelSet](#)
    - [HTS\\_Model](#)
    - [HTS\\_UttModel](#)
  - [Trees](#)
    - [HTS\\_Pattern](#)
    - [HTS\\_Question](#)
    - [HTS\\_Node](#)
    - [HTS\\_Tree](#)
    - [HTS\\_TreeSet](#)
  - [Parameter generation](#)
    - [HTS\\_DWin](#)
    - [HTS\\_SMatrices](#)
    - [HTS\\_PStream](#)
  - [Global settings](#)
    - [HTS\\_globalP](#)
    - [HTS\\_AudioSet](#)
    - [HTS\\_VocoderSet](#)
    - [HTS\\_Engine](#)
- [Functions](#)
  - [HTS\\_Engine functions](#)
    - [HTS\\_Engine\\_initialize](#)
    - [HTS\\_Engine\\_load\\_fn](#)
    - [HTS\\_Engine\\_load\\_fp](#)
    - [HTS\\_Engine\\_process](#)
    - [HTS\\_Engine\\_refresh](#)
    - [HTS\\_Engine\\_clear](#)
  - [HTS\\_Engine setting function](#)
    - [HTS\\_Engine\\_set\\_sampling\\_rate](#)
    - [HTS\\_Engine\\_set\\_fperiod](#)
    - [HTS\\_Engine\\_set\\_alpha](#)
    - [HTS\\_Engine\\_set\\_beta](#)
    - [HTS\\_Engine\\_set\\_rho](#)
    - [HTS\\_Engine\\_set\\_f0\\_std](#)
    - [HTS\\_Engine\\_set\\_f0\\_mean](#)
    - [HTS\\_Engine\\_set\\_uv](#)
    - [HTS\\_Engine\\_set\\_length](#)
    - [HTS\\_Engine\\_set\\_alignst](#)
    - [HTS\\_Engine\\_set\\_alignph](#)
    - [HTS\\_Engine\\_set\\_buff\\_size](#)
    - [HTS\\_Engine\\_set\\_stored\\_raw\\_data](#)
    - [HTS\\_Engine\\_get\\_sampling\\_rate](#)
    - [HTS\\_Engine\\_get\\_total\\_dur](#)
    - [HTS\\_Engine\\_get\\_total\\_frame](#)
    - [HTS\\_Engine\\_get\\_nsample](#)

- [HTS\\_Engine\\_get\\_pros\\_len](#)
- [HTS\\_Engine\\_get\\_pros](#)
- [HTS\\_Engine\\_get\\_stored\\_raw\\_data](#)
- [HTS\\_Model function](#)
  - [HTS\\_Model\\_load\\_from\\_labfp](#)
  - [HTS\\_Model\\_load\\_from\\_labfn](#)
  - [HTS\\_Model\\_load\\_from\\_string](#)
  - [HTS\\_Model\\_load\\_from\\_string\\_list](#)
- [HTS\\_Model setting function](#)
  - [HTS\\_Model\\_set\\_rate](#)
  - [HTS\\_Model\\_set\\_dur](#)
  - [HTS\\_Model\\_set\\_f0\\_level](#)
  - [HTS\\_Model\\_set\\_f0\\_range](#)
  - [HTS\\_Model\\_set\\_volume](#)
  - [HTS\\_Model\\_set\\_alpha](#)
- [HTS\\_VocoderSet function](#)
  - [HTS\\_VocoderSet\\_initialize](#)
  - [HTS\\_VocoderSet\\_synthesize](#)

## Preprocessor symbols for conditional compilation [↑](#)

- For embedded device

`HTS_EMBEDDED` (slightly faster setting is used)

- Audio device setting

`AUDIO_PLAY_WIN32` (for Windows 2000/XP/Vista C++ compiler)  
`AUDIO_PLAY_WINCE` (for Windows Mobile C++ compiler)  
`AUDIO_PLAY_NONE` (default)

- For [Festival speech synthesis system](#)

`FESTIVAL`

- Endian definition

`WORDS_BIGENDIAN` (e.g. PowerPC, Cell BE, SPARC, 680x0)  
`WORDS_LITTLEENDIAN` (e.g. x86, Alpha AXP)

[↑](#)

## Structures [↑](#)

### Models [↑](#)

#### `HTS_ModelSet?` [↑](#)

Set of HMMs and duration models.

<code>int nstate</code>	- # of HMM states
<code>int lf0stream</code>	- # of stream for F0
<code>int mcpvsize</code>	- vector size for spectrum
<code>int *nlf0pdf</code>	- # of PDFs at each state position (F0)
<code>int *nmcppdf</code>	- # of PDFs at each state position (spectrum)
<code>int ndurpdf</code>	- # of PDFs (duration)
<code>double **durpdf</code>	- array of PDFs (duration)
<code>double ***mcppdf</code>	- array of PDFs (spectrum)
<code>double ****lf0pdf</code>	- array of PDFs (F0)

```
double weight_interp - weight for model interpolation
```

↑

## HTS\_Model [↑](#)

A subword HMM in an utterance HMM.

char *name	- name of this HMM (name only)
char *lab	- label of this HMM (includes other information)
int durpdf	- duration PDF index
int *lf0pdf	- F0 PDF indexes
int *mcpdf	- spectrum PDF indexes
int *dur	- state durations (frame)
int totaldur	- total duration in this HMM (frame)
double **lf0mean	- mean vectors of F0 PDFs
double **lf0variance	- diag variances of F0 PDFs
double **mcpmean	- mean vectors of spectrum PDFs
double **mcpvariance	- diag variances of spectrum PDFs
HTS_Boolean *voiced	- voiced/unvoiced flags in this HMM
struct _HTS_Model *next	- pointer to the next subword HMM
HTS_Boolean bool_rate	- flag for speaking rate modification
HTS_Boolean bool_dur	- flag for duration modification
HTS_Boolean bool_f0_level	- flag for f0 level modification
HTS_Boolean bool_f0_range	- flag for f0 range modification
HTS_Boolean bool_volume	- flag for volume modification
HTS_Boolean bool_alpha	- flag for frequency warping modification
double lab_rate	- speaking rate specified in the given label
int lab_dur	- # of frames specified in the given label
double lab_f0_level	- f0 level specified in the given label
double lab_f0_range	- f0 range specified in the given label
double lab_volume	- volume specified in the given label
double lab_alpha	- frequency warping specified in the given label

↑

## HTS\_UttModel [?](#) [↑](#)

An utterance HMM.

HTS_Model *mhead	- list of subword HMMs (head)
HTS_Model *mtail	- list of subword HMMs (tail)
int nModel	- # of subword HMMs in this utterance HMM
int nState	- total # of HMM states in this utterance HMM
int totalframe	- total # of frames in this utterance

↑

## Trees [↑](#)

↑

## HTS\_Pattern [↑](#)

List of patterns in a question.

char *pat	- pattern string
struct _HTS_Pattern *next	- pointer to the next pattern

↑

## HTS\_Question [↑](#)

List of questions in HTS\_TreeSet[?](#).

char *qName	- name of this question
HTS_Pattern *phead	- list of patterns (head)
HTS_Pattern *ptail	- list of patterns (tail)
struct _HTS_Question *next	- pointer to the next question

↑

## HTS\_Node [↑](#)

List of tree nodes in a decision tree.

```
int idx           - index of this node
int pdf           - index of PDF for this node (leaf node only)
struct _HTS_Node *yes - pointer to its child node (yes)
struct _HTS_Node *no  - pointer to its child node (no)
struct _HTS_Node *next - pointer to the next node
HTS_Question *quest - question applied at this node
```

↑

## HTS\_Tree [↑](#)

List of decision trees in HTS\_TreeSet[?](#).

```
int state          - state position of this tree
HTS_Pattern *phead - list of patterns used in this tree (head)
HTS_Pattern *ptail  - list of patterns used in this tree (tail)
struct _HTS_Tree *next - pointer to the next tree
HTS_Node *root     - root node of this tree
HTS_Node *leaf      - list of leaf nodes in this tree
```

↑

## HTS\_TreeSet[?](#) [↑](#)

Set of decision trees.

```
HTS_Question *qhead[HTS_NUMMTYPE] - lists of questions for spectrum, F0 & duration (head)
HTS_Question *qtail[HTS_NUMMTYPE] - lists of questions for spectrum, F0 & duration (tail)
HTS_Tree *thead[HTS_NUMMTYPE]    - lists of trees for spectrum, F0 & duration (head)
HTS_Tree *ttail[HTS_NUMMTYPE]    - lists of trees for spectrum, F0 & duration (tail)
int nTrees[HTS_NUMMTYPE]         - # of trees for spectrum, F0 & duration
```

↑

## Parameter generation [↑](#)

↑

### HTS\_DWin [↑](#)

- Window coefficients to calculate dynamic features.

```
int num           - # of windows (static, delta, delta-delta -> 3)
int **width       - width of windows [0..num-1][0(left) 1(right)]
double **coef     - window coefficients [0..num-1][width[0]..width[1]]
int maxw[2]        - maximum width [0(left) 1(right)]
int max_L         - maximum width {maxw[0], maxw[1]}
```

↑

### HTS\_SMatrices [↑](#)

- Matrices/Vectors used in the speech parameter generation algorithm.

```
double **mseq   - mean vector sequence
double **ivseq  - inverse diag variance sequence
double *g        - vector used in the forward substitution
double **WUW    - W' U^-1 W
double *WUM     - W' U^-1 mu
```

↑

### HTS\_PStream [↑](#)

- PDF stream used in the speech parameter generation algorithm.

```
int vSize         - vector size of an observation vector (includes static & dynamic features)
int order         - vector size of static features
int T             - vector length (# of frames)
int width         - maximum width of dynamic feature windows
HTS_DWin dw       - dynamic feature windows
double **par      - output parameter vector
HTS_SMatrices sm - matrices/vectors for parameter generation
```

```
HTS_Boolean *voiced - voiced/unvoiced decision
```

## Global settings [↑](#)

### HTS\_globalP [↑](#)

- Global settings.

int rate	- sampling rate (Hz)
int fperiod	- frame shift (points)
double rho	- speaking rate
double alpha	- frequency warping
double beta	- postfiltering coefficient
double f0_std	- F0 multiply
double f0_mean	- F0 bias
double uv	- voiced/unvoiced threshold
double length	- total number of frames
HTS_Boolean algnst	- use state-level alignments from labels
HTS_Boolean algnph	- use phone-level alignments from labels
int totaldur	- total frame
int totalframe	- total frame
int nsample	- # of samples in a synthesized waveform
int buff_size	- buffer size of audio output device
short *raw_data	- synthesized waveform
HTS_Boolean stored_raw_data	- flag to store a synthesized waveform in raw_data

### HTS\_AudioSet? [↑](#)

- For MS Windows (Windows Mobile) audio output device.

HWAVEOUT hwaveout	- audio device handle
WAVEFORMATEX waveformatex	- wave formatex
short *buff	- current buffer
int buff_size	- current buffer size
int which_buff	- double buffering flag
HTS_Boolean now_buff_1	- double buffering flag
HTS_Boolean now_buff_2	- double buffering flag
WAVEHDR buff_1	- buffer
WAVEHDR buff_2	- buffer
int max_buff_size	- buffer size of audio output device

- For Linux, etc.

int i	- make compiler happy
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### HTS\_VocoderSet? [↑](#)

- MLSA filter settings. Usually you don't need to set this structure manually.

int fprd	- frame shift
int iprd	- interpolation period
int seed	- seed of random generator
int pd	- Pade approximation order (4 or 5)
unsigned long next	- temporary variable for random generator
HTS_Boolean gauss	- flag to use Gaussian noise
double p1	- used in excitation generation
double pc	- used in excitation generation
double pade[21]	- Pade coefficients
double *ppade	- Pade array
double *c, *cc, *cinc, *d1	- used in the MLSA filter
double rate	- sampling rate
int sw	- switch used in random generator
double r1, r2, s	- used in random generator
int x	- excitation signal
HTS_AudioSet *as	- pointer for audio device
int size	- buffer size for postfiltering
double *d	- used in postfiltering
double *g	- used in postfiltering

double *mc	- mel-cepstral coefficients
double *cep	- cepstral coefficients
double *ir	- impulse response
int o	- used in postfiltering
int irleng	- length of impulse response

↑

## HTS\_Engine [↑](#)

- HTS\_Engine itself.

HTS_ModelSet *ms	- sets of HMMs and duration models
HTS_TreeSet *ts	- sets of decision trees
HTS_PStream lf0pst	- PDF stream for F0
HTS_PStream mcppst	- PDF stream for spectrum
HTS_globalP gp	- global settings
int num_interp	- # of models for interpolation

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## Functions [↑](#)

### HTS\_Engine functions [↑](#)

`void HTS_Engine_initialize(HTS_Engine *engine);`

- Use: Initialize HTS\_Engine structure.
- Arguments:

HTS_Engine *engine	- Pointer for HTS_Engine structure
--------------------	------------------------------------

- **Attention!!:** To start HTS\_Engine module, first you should call this function.

↑

### HTS\_Engine\_load\_fn [↑](#)

`void HTS_Engine_load_fn(HTS_Engine *engine,
 char **fn_ms_lf0,char **fn_ms_mcp,char **fn_ms_dur,
 char **fn_ts_lf0,char **fn_ts_mcp,char **fn_ts_dur,
 int num_ws_lf0,char **fn_ws_lf0,
 int num_ws_mcp,char **fn_ws_mcp,
 double *rate_interp,int num_interp);`

- Use: Load models/trees/windows from files using given filenames.
- Arguments:

HTS_Engine *engine	- HTS_Engine structure
char **fn_ms_lf0	- F0 PDF file names
char **fn_ms_mcp	- spectrum PDF file names
char **fn_ms_dur	- duration PDF file names
char **fn_ts_lf0	- F0 tree file names
char **fn_ts_mcp	- spectrum tree file names
char **fn_ts_dur	- duration tree file names
int num_ws_lf0	- # of dynamic feature windows for F0
char **fn_ws_lf0	- dynamic feature window file names for F0
int num_ws_mcp	- # of dynamic feature windows for spectrum
char **fn_ws_mcp	- dynamic feature window file names for spectrum
double *rate_interp	- model interpolation rates
int num_interp	- # of models to be interpolated

- **Attention!!:** You should initialize variable *engine* using HTS\_Engine\_initialize before calling this function. If *rate\_interp==NULL*, interpolation rates of all models are set to the same value.

↑

### HTS\_Engine\_load\_fp [↑](#)

```

void HTS_Engine_load_fp(HTS_Engine *engine,
                        FILE **fp_ms_lf0,FILE **fp_ms_mcp,FILE **fp_ms_dur,
                        FILE **fp_ts_lf0,FILE **fp_ts_mcp,FILE **fp_ts_dur,
                        int num_ws_lf0,FILE **fp_ws_lf0,
                        int num_ws_mcp,FILE **fp_ws_mcp,
                        double *rate_interp,int num_interp);

```

- Use: Load models, trees & windows from files using given file pointers.
- Arguments:

HTS_Engine *engine	- HTS_Engine structure
FILE **fp_ms_lf0	- F0 PDF file pointers
FILE **fp_ms_mcp	- spectrum PDF file pointers
FILE **fp_ms_dur	- duration PDF file pointers
FILE **fp_ts_lf0	- F0 tree file pointers
FILE **fp_ts_mcp	- spectrum tree file pointers
FILE **fp_ts_dur	- duration tree file pointers
int num_ws_lf0	- # of dynamic feature windows for F0
FILE **fp_ws_lf0	- dynamic feature window file pointers for F0
int num_ws_mcp	- # of dynamic feature windows for spectrum
FILE **fp_ws_mcp	- dynamic feature window file pointers for spectrum
double *rate_interp	- model interpolation rates
int num_interp	- # of models to be interpolated

- **Attention!!:** You should initialize variable *engine* using HTS\_Engine\_initialize before calling this function. If rate\_interp==NULL, interpolation rates of all models are set to the same value.

## HTS\_Engine\_process [↑](#)

```

void HTS_Engine_process(HTS_Engine *engine,HTS_Model *model,
                       FILE *wavfp, FILE *rawfp, FILE *lf0fp,
                       FILE *mcpf, FILE *durfp, FILE *tracefp,
                       double *fr_f0, double *fr_power);

```

- Use: run HMM-based speech synthesis.
- Arguments:

HTS_Engine *engine	- HTS_Engine structure
HTS_Model *model	- HTS_Model structure
FILE *wavfp	- file pointer for RIFF waveform output
FILE *rawfp	- file pointer for raw audio output
FILE *lf0fp	- file pointer for generated (log) F0 sequence
FILE *mcpf	- file pointer for generated spectrum (mel-cepstrum)
FILE *durfp	- file pointer for predicted durations
FILE *tracefp	- file pointer for trace information
double *fr_f0	- Given F0 values
double *fr_power	- Given powers

- **Attention!!:** You can give F0 values or powers predicted by other modules via fr\_f0 and fr\_power.

## HTS\_Engine\_refresh [↑](#)

```

void HTS_Engine_refresh(HTS_Engine *engine, Model *mhead);

```

- Use: free model list.
- Arguments:

HTS_Engine *engine	- HTS_Engine structure
Model *mhead	- HTS_Model structure

- **Attention!!:** You should call this function if you want to run HTS\_Engine\_process multiple times.

## HTS\_Engine\_clear [↑](#)

```

void HTS_Engine_clear(HTS_Engine *engine);

```

- Use: free memory.

- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
```

## HTS\_Engine setting function [↑](#)

### HTS\_Engine\_set\_sampling\_rate [↑](#)

```
void HTS_Engine_set_sampling_rate(HTS_Engine *engine,int i);
```

- Use: set sampling frequency.
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
int          i      - sampling frequency (Hz), 0 < i <= 48000
```

- **Attention!!:** Default value is 16000.

### HTS\_Engine\_set\_fperiod [↑](#)

```
void HTS_Engine_set_fperiod(HTS_Engine *engine,int i);
```

- Use: set frame shift.
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
int          i      - frame shift (point), 0 < i <= 2000
```

- **Attention!!:** Default value is 80.

### HTS\_Engine\_set\_alpha [↑](#)

```
void HTS_Engine_set_alpha(HTS_Engine *engine,double f);
```

- Use: set frequency warping parameter alpha.
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
double        f      - alpha, 0.0 <= f <= 1.0
```

- **Attention!!:** Default value is 0.42.

### HTS\_Engine\_set\_beta [↑](#)

```
void HTS_SetBeta(HTS_Engine *engine,double f);
```

- Use: set postfiltering coefficient parameter beta.
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
double        f      - beta, -0.8 <= f <= 0.8
```

- **Attention!!:** Default value is 0.0. If you set beta large value, formant structure will be emphasized strongly.

### HTS\_Engine\_set\_rho [↑](#)

```
void HTS_Engine_set_rho(HTS_Engine *engine,double f);
```

- Use: set speaking rate control parameter rho.

- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
double      f      - rho, -1.0 <= f <= 1.0
```

- **Attention!!:** Default value is 0.0. If you set beta negative value, speaking rate of synthesized speech becomes fast.

## [HTS\\_Engine\\_set\\_f0\\_std](#) ↑

```
void HTS_Engine_set_f0_std(HTS_Engine *engine,double f);
```

- Use: set a parameter to be multiplied to generated F0 values
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
double      f      - F0 multiply value, 0.0 <= f <= 5.0
```

- **Attention!!:** Default value is 1.0.

## [HTS\\_Engine\\_set\\_f0\\_mean](#) ↑

```
void HTS_Engine_set_f0_mean(HTS_Engine *engine,double f);
```

- Use: set a parameter to be added to generated F0 values
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
double      f      - F0 bias value, 0.0 <= f <= 100.0
```

- **Attention!!:** Default value is 0.0.

## [HTS\\_Engine\\_set\\_uv](#) ↑

```
void HTS_Engine_set_uv(HTS_Engine *engine,double f);
```

- Use: set voiced/unvoiced threshold.
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
double      f      - voiced/unvoiced threshold, 0.0 <= f <= 1.0
```

- **Attention!!:** Default value 0.5.

## [HTS\\_Engine\\_set\\_length](#) ↑

```
void HTS_Engine_set_length(HTS_Engine *engine,double f);
```

- Use: set total length of utterance in second
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
double      f      - total length of utterance (second), 0.0 <= f <= 30.0
```

- **Attention!!:** Default value is 0.0 (using predicted durations by state duration models).

## [HTS\\_Engine\\_set\\_alignst](#) ↑

```
void HTS_Engine_set_alignst(HTS_Engine *engine,HTS_Boolean i);
```

- Use: set flag whether state-level alignments from given labels is used or not.
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure  
HTS_Boolean i - flag whether state-level alignments from given labels is used or not
```

- **Attention!!:** Default value is FALSE.

## HTS\_Engine\_set\_alignph [↑](#)

```
void HTS_Engine_set_alignph(HTS_Engine *engine,HTS_Boolean i);
```

- Use: set flag whether phone-level alignments from given labels is used or not.
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure  
HTS_Boolean i - flag whether phone-level alignments from given labels is used or not
```

- **Attention!!:** Default value is FALSE.

## HTS\_Engine\_set\_buff\_size [↑](#)

```
void HTS_Engine_set_buff_size(HTS_Engine *engine, int i);
```

- Use: set buffer size for audio device.
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure  
int i - buffer size (pt), 0 < i <= 48000
```

- **Attention!!:** Default value is 0. If i==0, direct audio play is turned off.

## HTS\_Engine\_set\_stored\_raw\_data [↑](#)

```
void HTS_Engine_set_stored_raw_data(HTS_Engine *engine,HTS_Boolean i);
```

- Use: set flag whether a synthesized waveform is stored in an array.
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure  
HTS_Boolean i - if TRUE, synthesized waveform is stored
```

- **Attention!!:** Default value is FALSE.

## HTS\_Engine\_get\_sampling\_rate [↑](#)

```
int HTS_Engine_get_sampling_rate(HTS_Engine *engine);
```

- Use: get sampling frequency.
- Return value: sampling frequency (Hz)
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
```

## HTS\_Engine\_get\_total\_dur [↑](#)

```
int HTS_Engine_get_total_dur(HTS_Engine *engine);
```

- Use: get total durations
- Return value: get total durations
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
```

## **HTS\_Engine\_get\_total\_frame** [↑](#)

```
int HTS_Engine_get_total_frame(HTS_Engine *engine);
```

- Use: get total # of frame.
- Return value: total # of frames.
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
```

[↑](#)

## **HTS\_Engine\_get\_nsample** [↑](#)

```
int HTS_Engine_get_nsample(HTS_Engine *engine);
```

- Use: get # of samples in a synthesized waveform.
- Return value: # of samples.
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
```

[↑](#)

## **HTS\_Engine\_get\_pros\_len** [↑](#)

```
int HTS_Engine_get_pros_len(HTS_Engine *engine);
```

- Use: get spectrum, F0 data length.
- Return value: data length.
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
```

[↑](#)

## **HTS\_Engine\_get\_pros** [↑](#)

```
HTS_Boolean HTS_Engine_get_pros(HTS_Engine *engine,int len,  
                                double *f0_data,double *power_data);
```

- Use: get mcp, If0 data array.
- Return value: if len==data length, return TRUE.
- Arguments:

```
HTS_Engine *engine      - HTS_Engine structure  
int          len        - data length  
double       *f0_data   - generated F0 sequence  
double       *power_data - generated spectrum (mel-cepstrum) sequence
```

[↑](#)

## **HTS\_Engine\_get\_stored\_raw\_data** [↑](#)

```
short *HTS_Engine_get_stored_raw_data(HTS_Engine *engine);
```

- Use: get stored raw data.
- Return value: short array which contains a synthesized waveform.
- Arguments:

```
HTS_Engine *engine - HTS_Engine structure
```

[↑](#)

## **HTS\_Model function** [↑](#)

[↑](#)

### **HTS\_Model\_load\_from\_labfp** [↑](#)

```
void HTS_Model_load_from_labfp(HTS_Model *mhead,FILE *labfp);
```

- Use: load model list from label file pointer.
- Arguments:

```
HTS_Model *mhead - HTS_Model structure pointer  
FILE      *labfp - label file pointer
```

↑

## HTS\_Model\_load\_from\_labfn [↑](#)

```
void HTS_Model_load_from_labfn (HTS_Model *mhead,char *fn);
```

- Use: load model list from label file name.
- Arguments:

```
HTS_Model *mhead - HTS_Model structure pointer  
char      *fn     - label file name
```

↑

## HTS\_Model\_load\_from\_string [↑](#)

```
void HTS_Model_load_from_string (HTS_Model *mhead,char *labdata);
```

- Use: load model list from string.
- Arguments:

```
HTS_Model *mhead - HTS_Model structure pointer  
char      *labdata - label string
```

↑

## HTS\_Model\_load\_from\_string\_list [↑](#)

```
void HTS_Model_load_from_string_list (HTS_Model *mhead,char **labdata,int size);
```

- Use: load model list from string list.
- Arguments:

```
HTS_Model *mhead - HTS_Model structure pointer  
char      **labdata - label string list  
int       size    - label string list size
```

↑

## HTS\_Model setting function [↑](#)

↑

### HTS\_Model\_set\_rate [↑](#)

```
void HTS_Model_set_rate(HTS_Model *m, int i, double f);
```

- Use: set speaking rate for the i-th subword HMM.
- Arguments:

```
HTS_Model *m - HTS_Model list  
int      i  - index of subword HMM to be used in HTS_Model list m  
double   f  - speaking rate
```

↑

### HTS\_Model\_set\_dur [↑](#)

```
void HTS_Model_set_dur(Model *m, int i, int d);
```

- Use: set phone-level duration for the i-th subword HMM.

- Arguments:

```
HTS_Model *m - HTS_Model list
int      i - index of subword HMM to be used in HTS_Model list m
int      d - # of frames (duration)
```

- **Attention!!:**  $0 < d$ .

↑

## HTS\_Model\_set\_f0\_level [↑](#)

```
void HTS_Model_set_f0_level (Model *m, int i, double f);
```

- Use: set F0 level for the i-th subword HMM.
- Arguments:

```
HTS_Model *m - HTS_Model list
int      i - index of subword HMM to be used in HTS_Model list m
double   f - F0 bias
```

- **Attention!!:**  $0.1 \leq f$ .

↑

## HTS\_Model\_set\_f0\_range [↑](#)

```
void HTS_Model_set_f0_range(Model *m, int i, double f);
```

- Use: set F0 range for the i-th subword HMM.
- Arguments:

```
HTS_Model *m - HTS_Model list
int      i - index of subword HMM to be used in HTS_Model list m
double   f - F0 range
```

- **Attention!!:**  $0.0 \leq f$ . Default value is 0.0.

↑

## HTS\_Model\_set\_volume [↑](#)

```
void HTS_Model_set_volume(Model *m, int i, double f);
```

- Use: set volume for the i-th subword HMM.
- Arguments:

```
HTS_Model *m - HTS_Model list
int      i - index of subword HMM to be used in HTS_Model list m
double   f - volume
```

- **Attention!!:**  $0.01 \leq f$ .

↑

## HTS\_Model\_set\_alpha [↑](#)

```
void HTS_Model_set_alpha(Model *m, int i, double f);
```

- Use: set frequency warping parameter alpha for the i-th subword HMM.
- Arguments:

```
HTS_Model *m - HTS_Model list
int      i - index of subword HMM to be used in HTS_Model list m
double   f - frequency warping parameter alpha
```

- **Attention!!:**  $0.0 \leq f$ .

↑

## HTS\_VocoderSet? function [↑](#)

↑

## HTS\_VocoderSet\_initialize [↑](#)

```
void HTS_VocoderSet_initialize (HTS_VocoderSet *vs, const int m,
                               const int rate, const int fperiod, int buff_size);
```

- Use: initialize the MLSA filter.
- Arguments:

HTS_VocoderSet *vs	- HTS_VocoderSet structure
const int m	- order of mel-cepstral coefficients
const int rate	- sampling frequency (Hz)
const int fperiod	- frame shift (point)
int buff_size	- buffer size for direct audio output

[↑](#)

## HTS\_VocoderSet\_synthesize [↑](#)

```
void HTS_VocoderSet_synthesize (HTS_VocoderSet *vs, const int m,
                                 double p, double *mc, double alpha, double beta,
                                 FILE *wavfp, FILE *rawfp, short *rawdata);
```

- Use: run the MLSA filter and synthesize waveform.
- Arguments:

HTS_VocoderSet *vs	- HTS_VocoderSet structure
const int m	- order of mel-cepstral coefficients
double p	- F0 value
double *mc	- mel-cepstral coefficients
double alpha	- frequency warping parameter alpha
double beta	- postfiltering parameter beta
FILE *wavfp	- file pointer to store synthesized waveform in RIFF format
FILE *rawfp	- file pointer to store synthesized waveform in raw audio