

# Examples for Using Speech Signal Processing Toolkit

## Ver. 3.5

SPTK working group

December 25, 2011

### Contents

<b>1</b>	<b>Basics</b>	<b>3</b>
1.1	Help message . . . . .	3
1.2	Data type conversion between “little endian” and “big endian.” . . . . .	3
1.3	Dump a binary data file . . . . .	3
1.4	Data type conversion from “short int” to “float” . . . . .	3
1.5	Plotting speech waveform on X-window . . . . .	3
1.6	Save the figure in an Encapsulated PostScript file . . . . .	4
1.7	Play a sound file . . . . .	4
1.8	Cut a portion out of a file . . . . .	4
<b>2</b>	<b>Pitch Extraction from Speech Waveform</b>	<b>5</b>
2.1	A pitch extractor . . . . .	5
2.2	Plotting the extracted pitch contour . . . . .	6
<b>3</b>	<b>Speech Analysis/Synthesis Based on Mel-Cepstral Representation</b>	<b>6</b>
3.1	Mel-cepstral analysis of speech . . . . .	6
3.2	Plotting spectral estimates from mel-cepstrum . . . . .	6
3.3	Plotting the spectral estimate with the FFT spectrum . . . . .	7
3.4	Speech synthesis from mel-cepstrum . . . . .	8
<b>4</b>	<b>Speech Analysis/Synthesis based on LPC</b>	<b>9</b>
4.1	LPC analysis of speech . . . . .	9
4.2	Plotting spectral estimates from LPC coefficients . . . . .	10
4.3	Plotting the spectral estimate with the FFT spectrum . . . . .	10
4.4	Speech synthesis from LPC coefficients . . . . .	11
4.5	Obtain PARCOR coefficients from LPC coefficients . . . . .	12
4.6	Speech synthesis from PARCOR coefficients . . . . .	12
4.7	Obtain LSP coefficients from LPC coefficients . . . . .	13
4.8	Speech synthesis from LSP coefficients . . . . .	13
<b>5</b>	<b>Speech Analysis/Synthesis Based on Mel-Generalized Cepstral Representation</b>	<b>14</b>
5.1	Mel-generalized cepstral analysis of speech . . . . .	14
5.2	Plotting spectral estimates from mel-generalized cepstrum . . . . .	15
5.3	Plotting the spectral estimate with the FFT spectrum . . . . .	16
5.4	Speech synthesis from mel-generalized cepstrum . . . . .	17

<b>6</b>	<b>Vector Quantization of Mel-Cepstrum</b>	<b>18</b>
6.1	Train a (very small) Codebook . . . . .	18
6.2	Encode (training vectors) . . . . .	18
6.3	Decode (training vectors) . . . . .	19
6.4	Plotting original and quantized spectra . . . . .	19
6.5	Performance evaluation on the training data . . . . .	20
6.6	Speech synthesis from quantized mel-cepstrum . . . . .	21
<b>7</b>	<b>Preparation of Speech Parameter for Speech Recognition</b>	<b>22</b>
7.1	Cepstrum derived from LPC analysis (LPC cepstrum) . . . . .	22
7.2	Mel-cepstrum derived from LPC analysis (LPC mel-cepstrum) . . . . .	22
7.3	Mel-cepstrum obtained by mel-cepstral analysis . . . . .	22
7.4	Mel-cepstrum derived from mel-generalized cepstral analysis . . . . .	23
7.5	Plotting spectra for each speech recognition parameter . . . . .	23
<b>8</b>	<b>Playing with the Vocoder Based on Mel-Cepstrum</b>	<b>24</b>
8.1	High- or low-pitched voice . . . . .	24
8.2	Fast- or slow-speaking voice . . . . .	25
8.3	Hoarse voice . . . . .	25
8.4	Robotic voice . . . . .	25
8.5	Child-like or deep voice . . . . .	25
8.6	Various voices . . . . .	25
<b>9</b>	<b>Speech Synthesis Based on HMM</b>	<b>26</b>
9.1	Speech parameter generation from a sequence of HMMs . . . . .	26
9.2	Plotting spectra calculated from generated mel-cepstrum . . . . .	26
9.3	Speech synthesis from the generated mel-cepstrum . . . . .	26
9.4	Check the given mean and variance vectors . . . . .	27
9.4.1	Dump static feature vectors . . . . .	27
9.4.2	Dump variance vectors of static feature vectors . . . . .	28
9.4.3	Dump dynamic feature vectors (delta) . . . . .	28
9.4.4	Dump variance vectors of dynamic feature vectors (delta) . . . . .	28
9.5	Speech synthesis without dynamic feature . . . . .	28

# 1 Basics

## 1.1 Help message

```
impulse -h
```

## 1.2 Data type conversion between “little endian” and “big endian.”

**Files:** data.short: speech data included in this example (short integer, 16 kHz sampling, little endian)  
data.short-b: speech data (short integer, 16 kHz sampling, big endian)

```
swab +s < data.short > data.short-b
```

## 1.3 Dump a binary data file

**Files:** data.short: speech data included in this example (short integer, 16 kHz sampling)

```
dmp +s data.short | less
```

## 1.4 Data type conversion from “short int” to “float”

**Files:** data.short: speech data included in this example (short integer, 16 kHz sampling)  
[data.float](#): speech data (float, 16 kHz sampling)<sup>1,2</sup>

```
x2x +sf < data.short > data.float
```

## 1.5 Plotting speech waveform on X-window

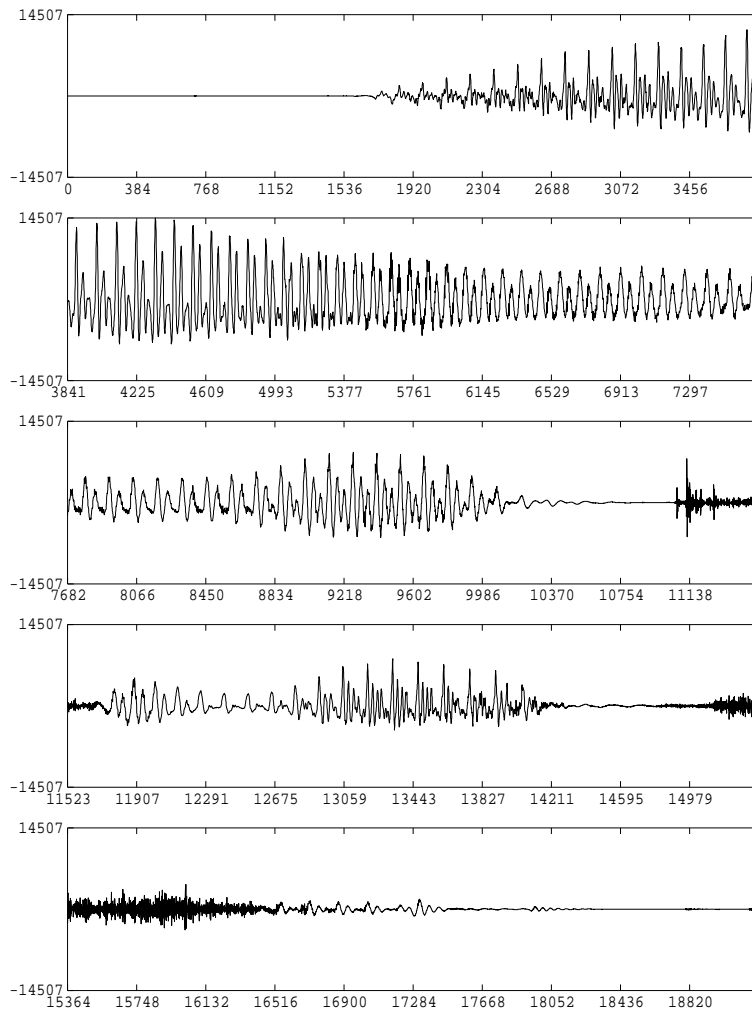
**Files:** data.short: speech data included in this example (short integer, 16 kHz sampling)

```
gwave +s data.short | xgr
```

---

<sup>1</sup>By clicking links in this PDF file, your PC may play some speech files, which were converted from “float” format into “wav” format (16 kHz sampling, 16-bit integer).

<sup>2</sup>If you compiled SPTK with “--enable-double” option, please use “+sd” option instead of “+sf” and “+d” option instead of “+f”.



## 1.6 Save the figure in an Encapsulated PostScript file

**Files:** data.short: speech data included in this example (short integer, 16 kHz sampling)  
figure.eps: Encapsulated PostScript file

```
gwave +s data.short | psgr > figure.eps
```

## 1.7 Play a sound file

**Files:** data.short: speech data included in this example (short integer, 16 kHz sampling)

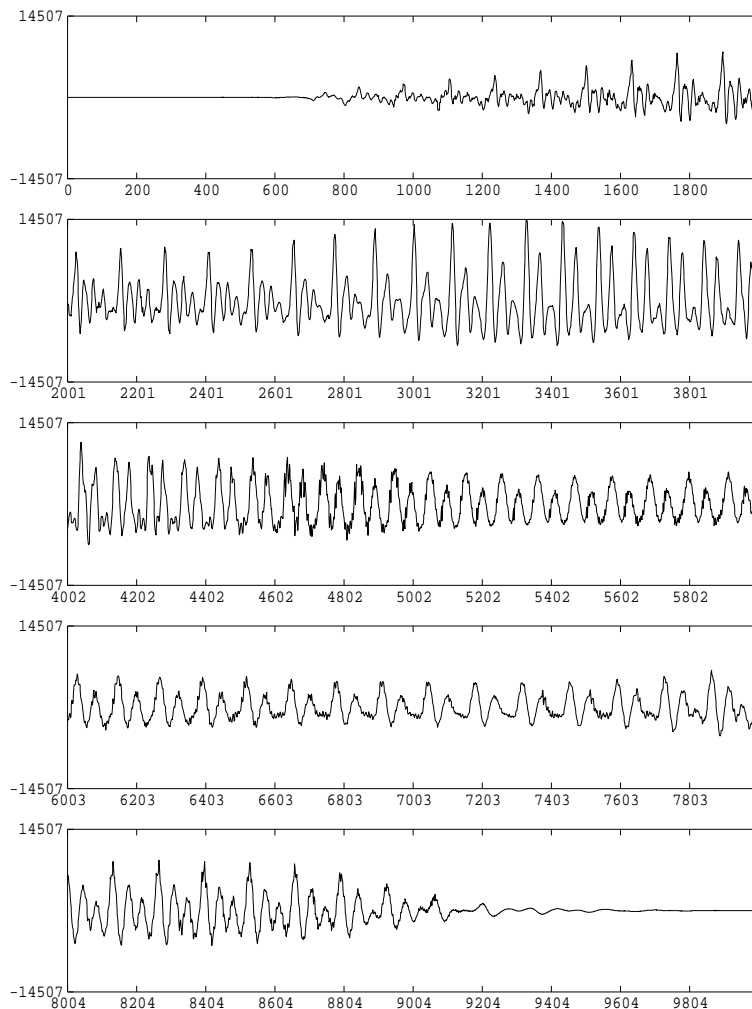
**Note:** This works only on Linux, Solaris, and FreeBSD.

```
da +s -s 16 -a 100 data.short
```

## 1.8 Cut a portion out of a file

**Files:** data.short: speech data included in this example (short integer, 16 kHz sampling)

```
bcut +s -s 1000 -e 11000 < data.short |\
gwave +s | xgr
```



## 2 Pitch Extraction from Speech Waveform

### 2.1 A pitch extractor

**Files:** data.short: speech data included in this example (short integer, 16 kHz sampling)

**Conditions:** frame period: 80 points (5 ms)  
minimum fundamental frequency for search: 80 Hz  
maximum fundamental frequency for search: 165 Hz

**Note:** Options should be adjusted for each speech data.

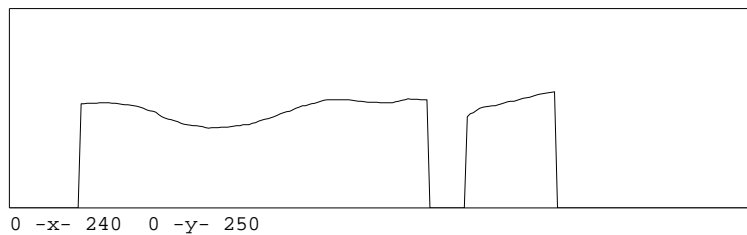
```
x2x +sf data.short | pitch -a 1 -s 16 -p 80 -L 80 -H 165 > data.pitch
```

## 2.2 Plotting the extracted pitch contour

**Files:** data.pitch: pitch data extracted from speech data "data.short"

**Conditions:** Minimum value of vertical axis: 0.0  
Maximum value of vertical axis: 250.0  
Width: 15 cm  
Height: 4 cm

```
fdrw -y 0 250 -W 1.5 -H 0.4 < data.pitch | xgr
```



## 3 Speech Analysis/Synthesis Based on Mel-Cepstral Representation

### 3.1 Mel-cepstral analysis of speech

**Files:** data.short: speech data included in this example (short integer, 16 kHz sampling)  
data.mcep: mel-cepstrum (float)

**Conditions:** frame length: 400 points (25 ms)  
frame period: 80 points (5 ms)  
window: Blackman window  
analysis order: 20  
frequency warping parameter:  $\alpha = 0.42$   
FFT size: 512 points

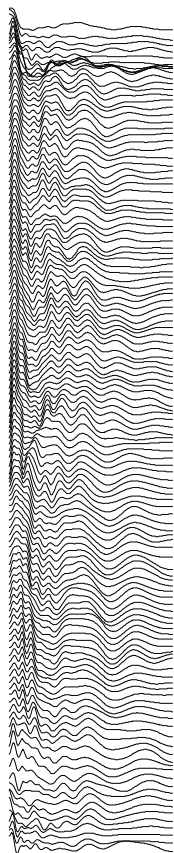
```
x2x +sf < data.short | frame -l 400 -p 80 | window -l 400 -L 512 |\
mcep -l 512 -m 20 -a 0.42 > data.mcep
```

### 3.2 Plotting spectral estimates from mel-cepstrum

**Files:** data.mcep: mel-cepstrum (float)

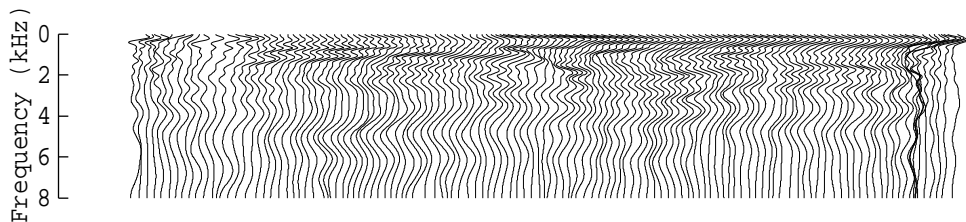
**Conditions:** analysis order: 20  
frequency warping parameter:  $\alpha = 0.42$   
FFT size: 512 points  
plotted frames: from 10-th to 135-th  
sampling frequency: 16 kHz

```
bcut +f -n 20 -s 10 -e 135 < data.mcep |\
mgc2sp -m 20 -a 0.42 -g 0 -l 512 | grlogsp -l 512 -x 8 | xgr
```



0 2 4 6 8  
Frequency (kHz)

```
bcut +f -n 20 -s 10 -e 135 < data.mcep |\
mgc2sp -m 20 -a 0.42 -g 0 -l 512 | grlogsp -l 512 -x 8 -t | xgr
```



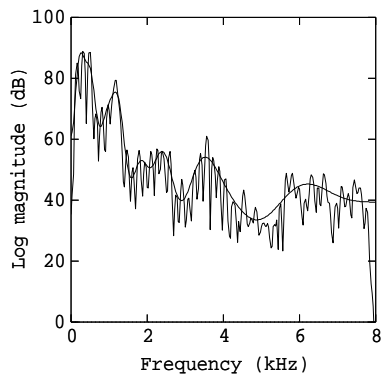
### 3.3 Plotting the spectral estimate with the FFT spectrum

**Files:** data.mcep: mel-cepstrum (float)

**Conditions:** analysis order: 20  
frequency warping parameter:  $\alpha = 0.42$   
FFT size: 512 points

plotted frame: 65-th  
sampling frequency: 16 kHz

```
( x2x +sf < data.short | frame -l 400 -p 80 | \  
bcut +f -l 400 -s 65 -e 65 |\  
window -l 400 -L 512 | spec -l 512 |\  
glogsp -l 512 -x 8 -p 2 ;\  
\  
bcut +f -n 20 -s 65 -e 65 < data.mcep |\  
mgc2sp -m 20 -a 0.42 -g 0 -l 512 | glogsp -l 512 -x 8 ) | xgr
```



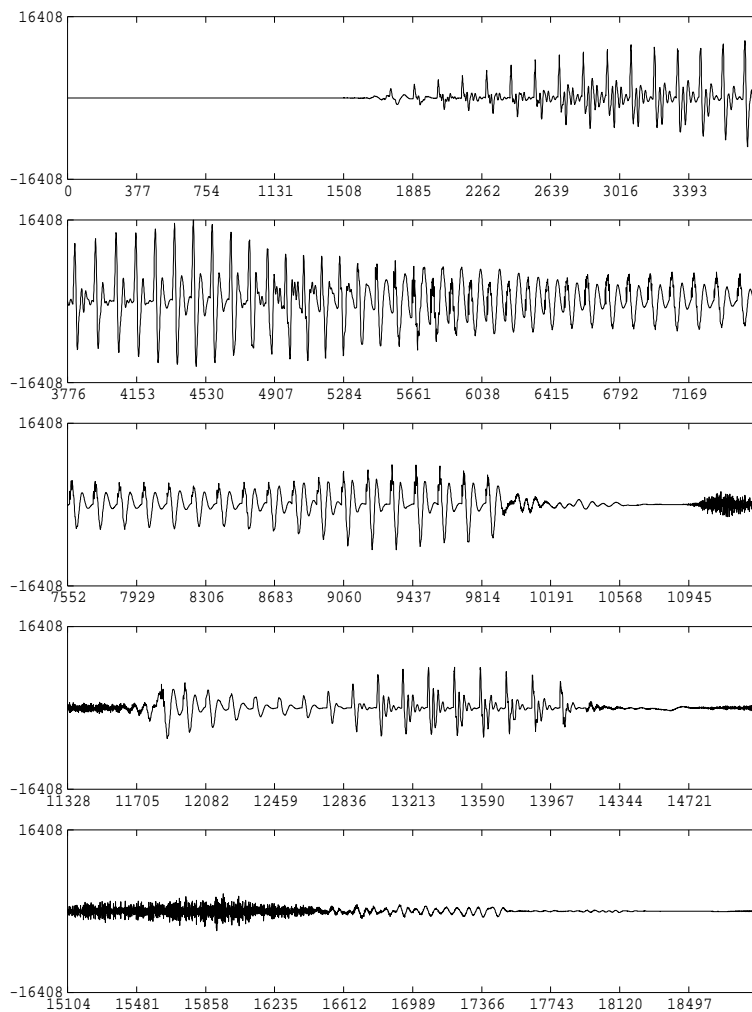
### 3.4 Speech synthesis from mel-cepstrum

**Files:** data.pitch: pitch data extracted from speech data "data.short"  
data.mcep: mel-cepstrum (float)  
[data.mcep.syn](#): synthesized speech (float)

**Conditions:** frame period: 80 points (5 ms)  
analysis order: 20  
frequency warping parameter:  $\alpha = 0.42$

```
excite -p 80 data.pitch |\  
mlsadf -m 20 -a 0.42 -p 80 data.mcep > data.mcep.syn  
  
gwave data.mcep.syn | xgr
```





```
da +f -s 16 data.mcep.syn
```

## 4 Speech Analysis/Synthesis based on LPC

### 4.1 LPC analysis of speech

**Files:** data.short: speech data included in this example (short integer, 16 kHz sampling)  
 data.lpc: LPC coefficients (float)

**Conditions:** frame length: 400 points (25 ms)  
 frame period: 80 points (5 ms)  
 window: Blackman window  
 analysis order: 20

```
x2x +sf < data.short | frame -l 400 -p 80 | window -l 400 |\
lpc -l 400 -m 20 > data.lpc
```

## 4.2 Plotting spectral estimates from LPC coefficients

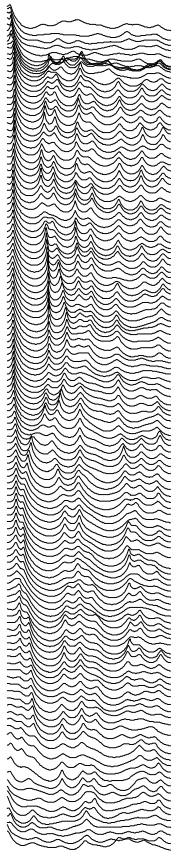
**Files:** data.lpc: LPC coefficients (float)

**Conditions:** analysis order: 20

```
bcut +f -n 20 -s 10 -e 135 < data.lpc |\
spec -l 512 -n 20 | grlogsp -l 512 -x 8 | xgr
```

or

```
bcut +f -n 20 -s 10 -e 135 < data.lpc |\
mgc2sp -m 20 -a 0 -g -1 -n -u -l 512 |\
grlogsp -l 512 -x 8 | xgr
```



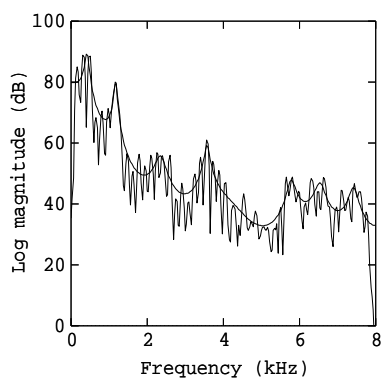
0 2 4 6 8  
Frequency (kHz)

## 4.3 Plotting the spectral estimate with the FFT spectrum

**Files:** data.lpc: LPC coefficients (float)

**Conditions:** analysis order: 20  
plotted frame: 65-th  
sampling frequency: 16 kHz

```
( x2x +sf < data.short | frame -l 400 -p 80 | \  
bcut +f -l 400 -s 65 -e 65 |\  
window -l 400 -L 512 | spec -l 512 |\  
glogsp -l 512 -x 8 -p 2 ;\  
\  
bcut +f -n 20 -s 65 -e 65 < data.lpc > data.tmp ;\  
spec -l 512 -n 20 -p data.tmp | glogsp -l 512 -x 8 ;\  
\rm data.tmp ) | xgr
```

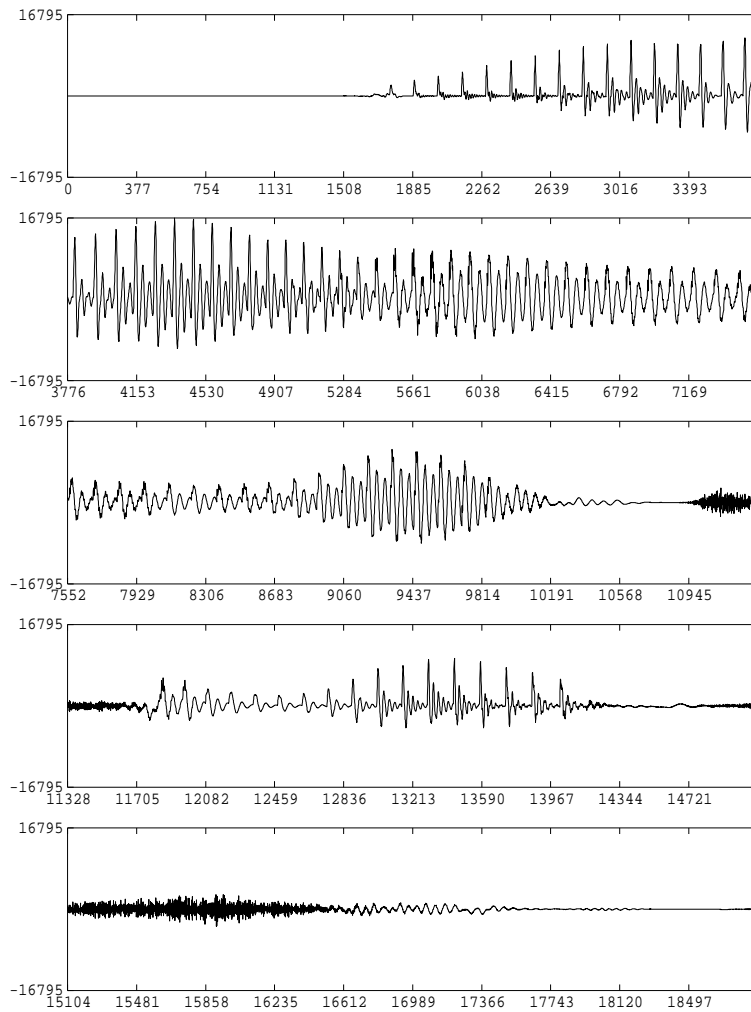


#### 4.4 Speech synthesis from LPC coefficients

**Files:** data.pitch: pitch data extracted from speech data "data"  
data.lpc: LPC coefficients (float)  
[data.lpc.syn](#): synthesized speech (float)

**Conditions:** frame period: 80 points (5 ms)  
analysis order: 20

```
excite -p 80 data.pitch | poledf -m 20 -p 80 data.lpc > data.lpc.syn  
gwave +f data.lpc.syn | xgr
```



```
da +f -s 16 data.lpc.syn
```

#### 4.5 Obtain PARCOR coefficients from LPC coefficients

**Files:** data.lpc: LPC coefficients (float)  
 data.par: PARCOR coefficients (float)

**Conditions:** analysis order: 20

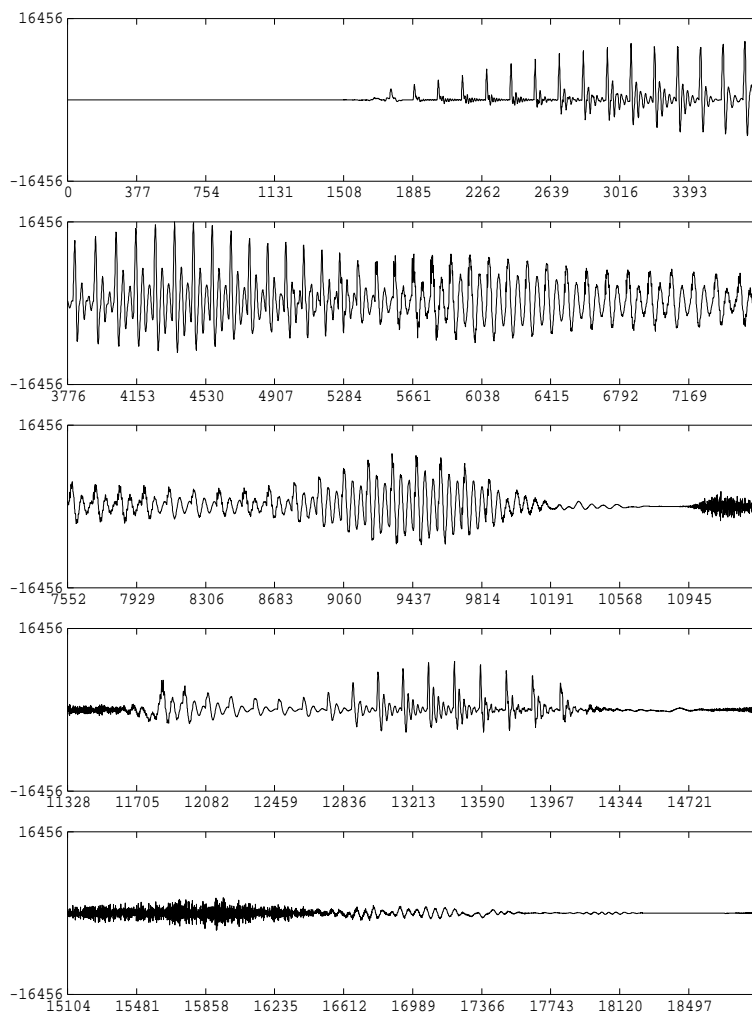
```
lpc2par -m 20 < data.lpc > data.par
```

#### 4.6 Speech synthesis from PARCOR coefficients

**Files:** data.pitch: pitch data extracted from speech data "data" (float)  
 data.par: PARCOR coefficients (float)  
[data.par.syn](#): synthesized speech (float)

**Conditions:** frame period: 80 points (5 ms)  
 analysis order: 20

```
excite -p 80 data.pitch | ltcdf -m 20 -p 80 data.par > data.par.syn
gwave +f data.par.syn | xgr
```



#### 4.7 Obtain LSP coefficients from LPC coefficients

**Files:** data.lpc: LPC coefficients (float)  
 data.lsp: LSP coefficients (float)

**Conditions:** analysis order: 20  
 split number of unit circle: 256

```
lpc2lsp -m 20 -n 256 < data.lpc > data.lsp
```

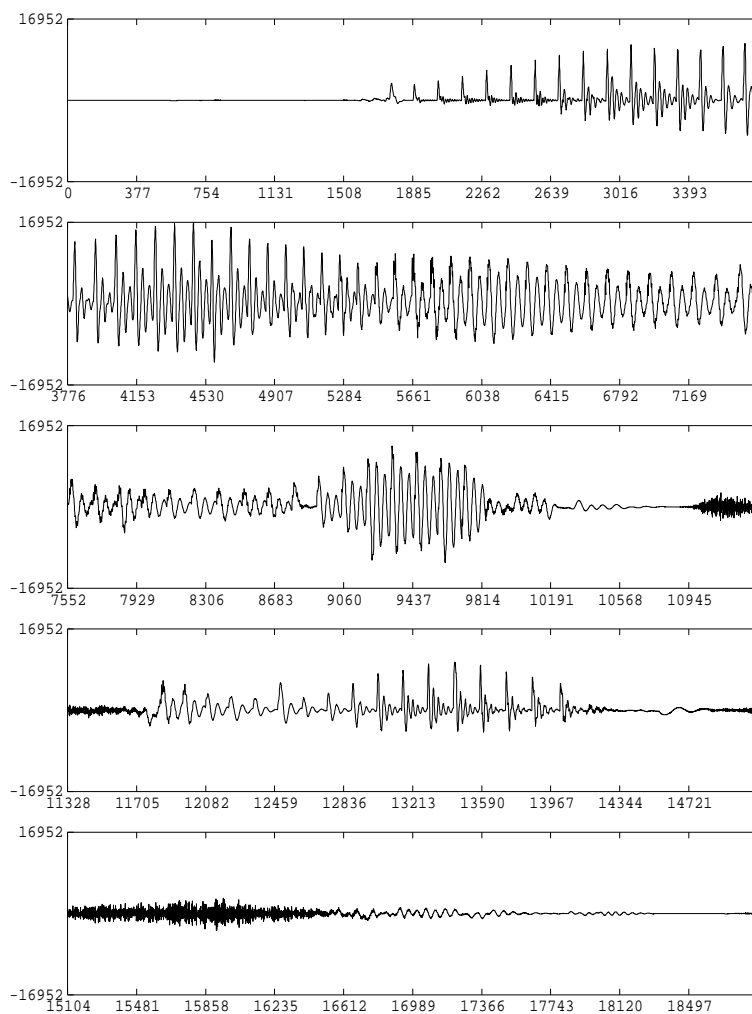
#### 4.8 Speech synthesis from LSP coefficients

**Files:** data.pitch: pitch data extracted from speech data "data"  
 data.lsp: LSP coefficients (float)  
[data.lsp.syn](#): synthesize speech (float)

**Conditions:** frame period: 80 points (5 ms)  
analysis order: 20

```
excite -p 80 data.pitch | lspdf -m 20 -p 80 data.lsp > data.lsp.syn
```

```
gwave +f data.lsp.syn | xgr
```



```
da +f -s 16 data.lsp.syn
```

## 5 Speech Analysis/Synthesis Based on Mel-Generalized Cepstral Representation

### 5.1 Mel-generalized cepstral analysis of speech

**Files:** speech data: data (short integer, 10 kHz sampling)  
data.mgcep: mel-generalized cepstrum (float)

**Conditions:** frame length: 400 points (25 ms)  
frame period: 80 points (5 ms)  
window: Blackman window  
analysis order: 20  
frequency warping parameter:  $\alpha = 0.42$   
power parameter:  $\gamma = -1/2$

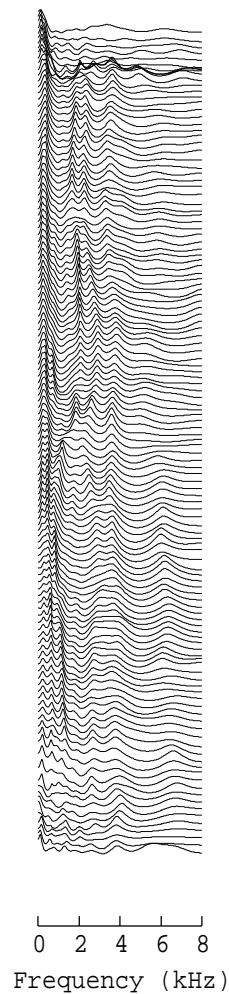
```
x2x +sf < data.short | frame -l 400 -p 80 | window -l 400 -L 512 |\
mgcep -m 20 -a 0.42 -c 2 -l 512 > data.mgcep
```

## 5.2 Plotting spectral estimates from mel-generalized cepstrum

**Files:** data.mgcep: mel-generalize cepstrum (float)

**Conditions:** analysis order: 20  
frequency warping parameter:  $\alpha = 0.42$   
power parameter:  $\gamma = -1/2$   
plotted frames: from 10-th to 135-th  
sampling frequency: 16 kHz

```
bcut +f -n 20 -s 10 -e 135 < data.mgcep |\
mgc2sp -m 20 -a 0.42 -c 2 -l 512 | grlogsp -l 512 -x 8 | xgr
```



### 5.3 Plotting the spectral estimate with the FFT spectrum

**Files:** data.mgcep: mel-generalized cepstrum (float)

**Conditions:** analysis order: 20  
frequency warping parameter:  $\alpha = 0.42$   
power parameter:  $\gamma = -1/2$   
FFT size: 512 points  
plotted frame: 65-th  
sampling frequency: 16 kHz

```
( x2x +sf < data.short | frame -l 400 -p 80 | \  

bcut +f -l 400 -s 65 -e 65 |\  

window -l 400 -L 512 | spec -l 512 |\  

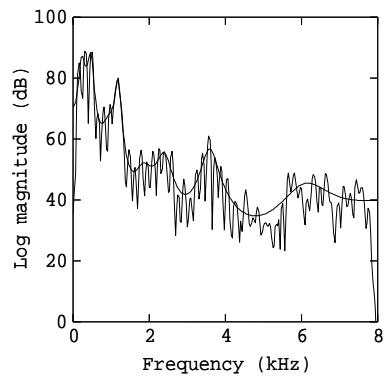
glogsp -l 512 -x 8 -p 2 ;\  

\  

bcut +f -n 20 -s 65 -e 65 < data.mgcep |\  

mgc2sp -m 20 -a 0.42 -c 2 -l 512 | glogsp -l 512 -x 8 ) | xgr
```





## 5.4 Speech synthesis from mel-generalized cepstrum

**Files:** data.pitch: pitch data extracted from speech data "data"

data.mgcep: mel-generalized cepstrum (float)

[data.mgcep.syn](#): synthesized speech (float)

**Conditions:** frame period: 80 points (5 ms)

analysis order: 20

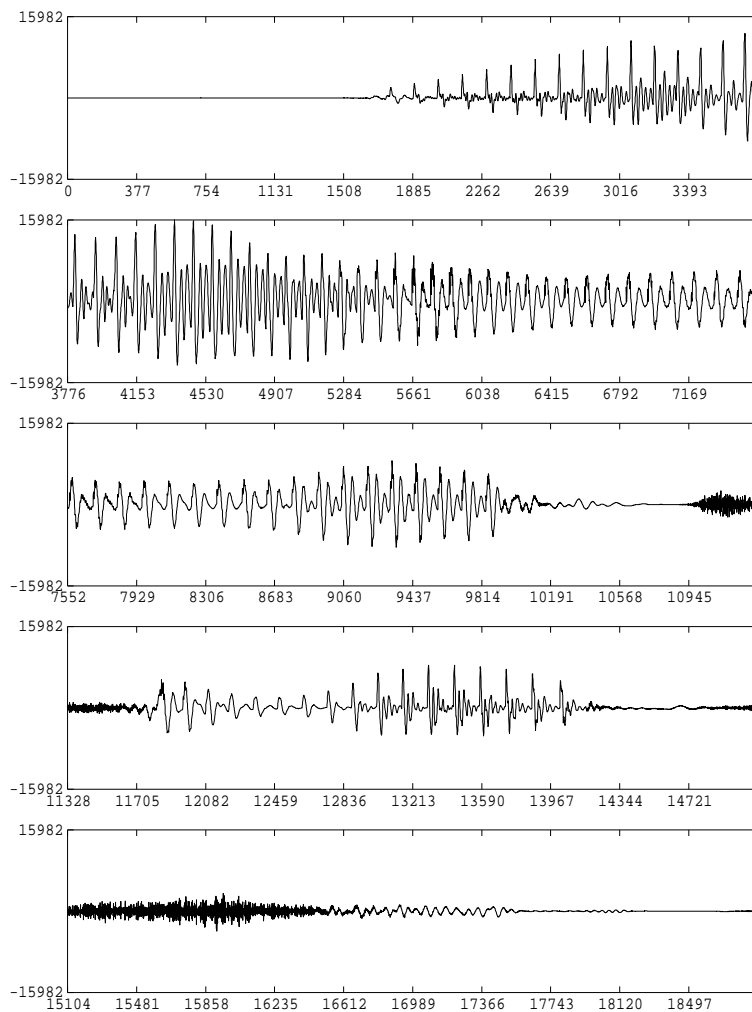
frequency warping parameter:  $\alpha = 0.42$

power parameter:  $\gamma = -1/2$

```
excite -p 80 data.pitch |\
```

```
mglsadf -m 20 -a 0.42 -c 2 -p 80 data.mgcep > data.mgcep.syn
```

```
gwave +f data.mgcep.syn | xgr
```



da +f -s 16 data.mgcep.syn

## 6 Vector Quantization of Mel-Cepstrum

### 6.1 Train a (very small) Codebook

**Files:** data.mcep: mel-cepstrum for training (float)  
codebook.mcep: codebook (float)

**Conditions:** vector size: 21 (analysis order: 20)  
codebook size: 32

lbg -n 20 -e 32 < data.mcep > codebook.mcep

### 6.2 Encode (training vectors)

**Files:** codebook.mcep: codebook (float)  
data.mcep.index: index (int)

**Conditions:** vector size: 21 (analysis order: 20)  
codebook size: 32

```
vq -n 20 codebook.mcep < data.mcep > data.mcep.index
```

### 6.3 Decode (training vectors)

**Files:** codebook.mcep: codebook (float)  
data.mcep.index: index (int)  
data.mcep.vq: quantized mel-cepstrum (float)

**Conditions:** vector size: 21 (analysis order: 20)  
codebook size: 32

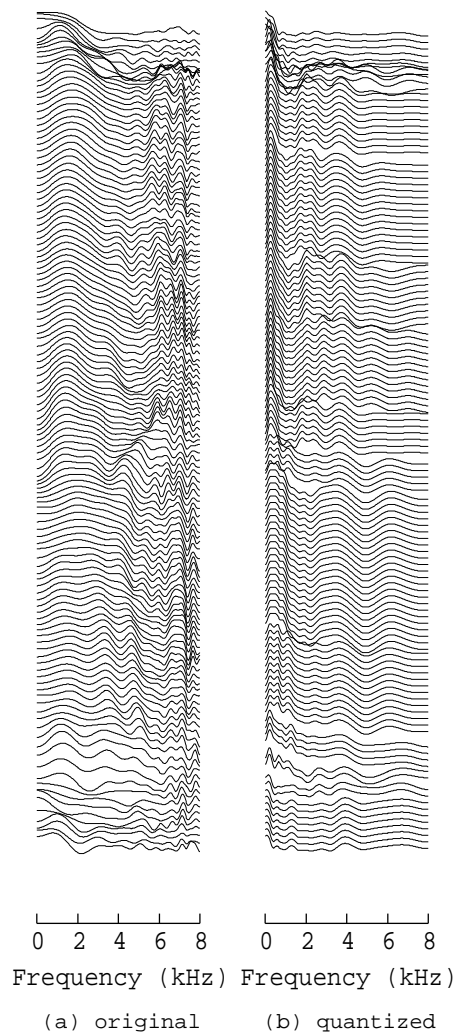
```
ivq -n 20 codebook.mcep < data.mcep.index > data.mcep.vq
```

### 6.4 Plotting original and quantized spectra

**Files:** data.mcep: original mel-cepstrum (float)  
data.mcep.vq: quantized mel-cepstrum (float)

**Conditions:** analysis order: 20  
frequency warping parameter:  $\alpha = 0.42$   
plotted frames: from 10-th to 135-th  
sampling frequency: 16 kHz

```
( bcut +f -n 20 -s 10 -e 135 < data.mcep |\
mgc2sp -m 20 -a 0.42 -g 0 -l 512 |\
grlogsp -l 512 -x 8 -o 1 -c "(a) original" ;\
\
bcut +f -n 20 -s 10 -e 135 < data.mcep.vq |\
mgc2sp -m 20 -a 0.42 -g 0 -l 512 |\
grlogsp -l 512 -x 8 -o 2 -c "(b) quantized" ) | xgr
```



## 6.5 Performance evaluation on the training data

**Files:** codebook.mcep: codebook (float)  
 data.mcep.index: index (int)  
 data.mcep.vq: quantized vectors (float)  
 data.mcep.vq.cdists: cepstrum distortion in dB (float)

**Conditions:** vector size: 21 (analysis order: 20)  
 codebook size: 32

```
freqt -a 0.42 -m 20 -A 0 -M 255 < data.mcep > data.mcep.cep
freqt -a 0.42 -m 20 -A 0 -M 255 < data.mcep.vq | \
cdist data.mcep.cep -m 255 > data.mcep.vq.cdists
\rm data.mcep.cep
```

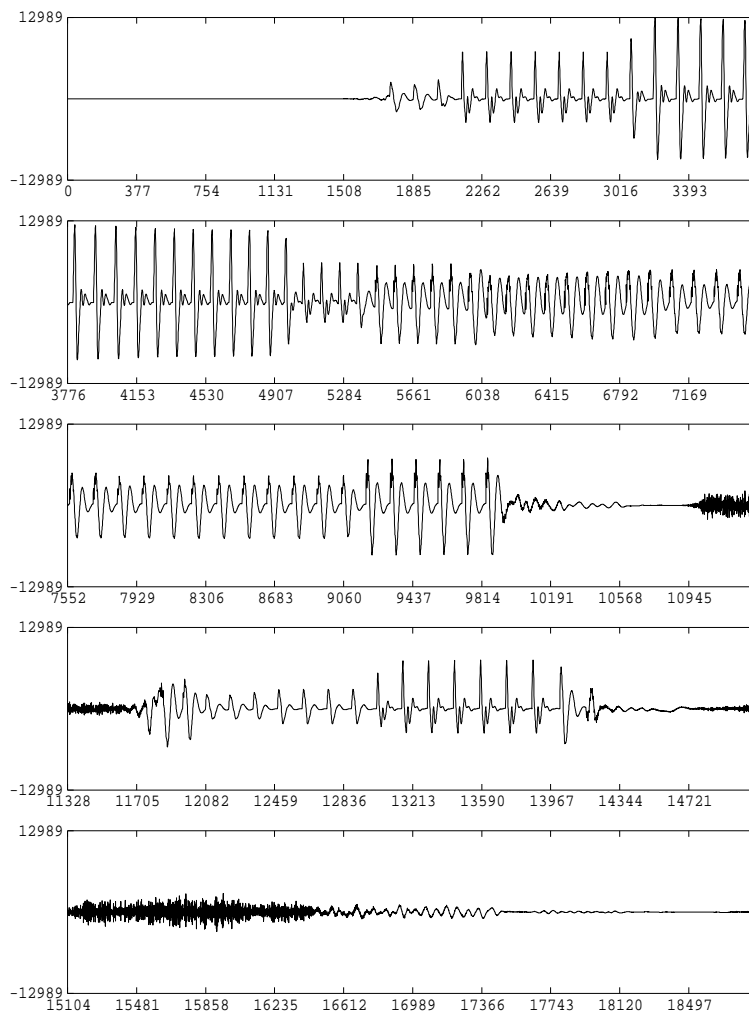
## 6.6 Speech synthesis from quantized mel-cepstrum

**Files:** data.pitch: pitch data extracted from speech data "data.short"  
data.mcep.vq: quantized mel-cepstrum (float)  
[data.mcep.vq.syn](#): synthesized speech (float)

**Conditions:** frame period: 80 points (5 ms)  
analysis order: 20  
frequency warping parameter:  $\alpha = 0.42$

```
excite -p 80 data.pitch |\
mlsadf -m 20 -a 0.42 -p 80 data.mcep.vq > data.mcep.vq.syn

gwave +f data.mcep.vq.syn | xgr
```



```
da +f -s 16 data.mcep.vq.syn
```

## 7 Preparation of Speech Parameter for Speech Recognition

### 7.1 Cepstrum derived from LPC analysis (LPC cepstrum)

**Files:** data.short: speech data included in this example (short integer, 16 kHz sampling)

**Conditions:** frame length: 400 points (25 ms)  
frame period: 80 points (5 ms)  
window: Blackman window  
analysis order: 12  
order of LPC cepstrum: 12

```
x2x +sf < data.short | frame -l 400 -p 80 | window -l 400 |\
lpc -l 400 -m 12 | lpc2c -m 12 -M 12 > data.lpc.cep
```

### 7.2 Mel-cepstrum derived from LPC analysis (LPC mel-cepstrum)

**Files:** data.short: speech data included in this example (short integer, 16 kHz sampling)

**Conditions:** frame length: 400 points (25 ms)  
frame period: 80 points (5 ms)  
window: Blackman window  
analysis order: 12  
order of LPC mel-cepstrum: 12

```
x2x +sf < data.short | frame -l 400 -p 80 | window -l 400 |\
lpc -l 400 -m 12 |\
lpc2c -m 12 -M 256 |\
freqt -m 256 -a 0 -M 12 -A 0.42 > data.lpc.mcep
```

or

```
x2x +sf < data.short | frame -l 400 -p 80 | window -l 400 |\
lpc -l 400 -m 12 |\
mgc2mgc -m 12 -a 0 -g -1 -n -u -M 12 -A 0.42 -G 0 > data.lpc.mcep
```

### 7.3 Mel-cepstrum obtained by mel-cepstral analysis

**Files:** data.short: speech data included in this example (short integer, 16 kHz sampling)  
data.mcep: mel-cepstrum (float)

**Conditions:** frame length: 400 points (25 ms)  
frame period: 80 points (5 ms)  
window: Blackman window  
analysis order: 20  
frequency warping parameter:  $\alpha = 0.42$   
FFT size: 512 points

```
x2x +sf < data.short | frame -l 400 -p 80 | window -l 400 -L 512 |\
mcep -l 512 -m 12 -a 0.42 > data.mcep.mcep
```

## 7.4 Mel-cepstrum derived from mel-generalized cepstral analysis

**Files:** data.short: speech data included in this example (short integer, 10 kHz sampling)

**Conditions:** frame length: 400 points (25 ms)  
frame period: 80 points (5 ms)  
Blackman window  
FFT size: 512 points  
( $\alpha, \gamma$ ) for analysis: (0.42, -0.5)  
analysis order: 12  
order of mel-cepstrum: 12

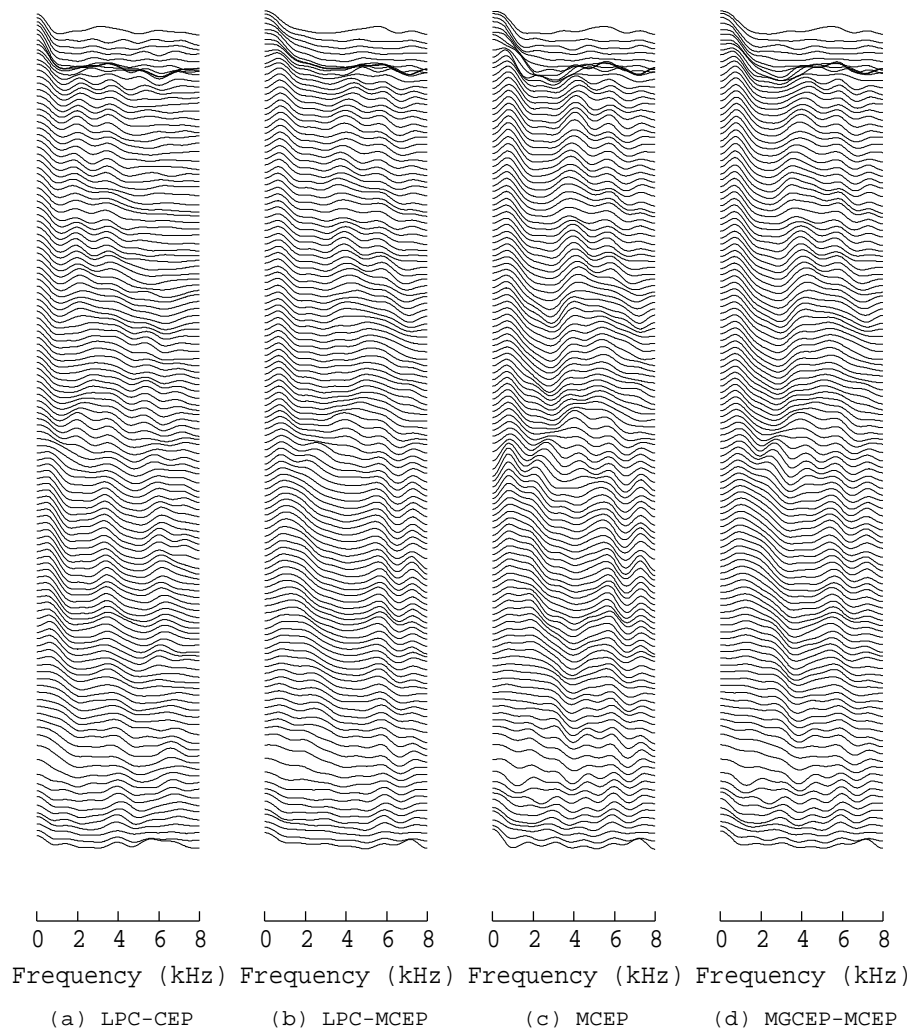
```
x2x +sf < data.short | frame -l 400 -p 80 | window -l 400 -L 512 |\
mgcep -m 12 -a 0.42 -c 2 -l 512 |\
mgc2mgc -m 12 -a 0.42 -c 2 -M 12 -A 0.42 -G 0 > data.mgcep.mcep
```

## 7.5 Plotting spectra for each speech recognition parameter

**Files:** data.lpc.cep: LPC cepstrum (float)  
data.lpc.mcep: LPC mel-cepstrum (float)  
data.mcep.mcep: mel-cepstrum (float)  
data.mgcep.mcep: mel-cepstrum derived from mel-generalized cepstrum (float)

**Conditions:** plotted frames: from 10-th to 135-th

```
\
bcut +f -n 12 -s 10 -e 135 < data.lpc.cep |\
mgc2sp -m 12 -a 0 -g 0 -l 512 |\
grlogsp -l 512 -x 8 -O 1 -c "(a) LPC-CEP" ;\
\
bcut +f -n 12 -s 10 -e 135 < data.lpc.mcep |\
mgc2sp -m 12 -a 0.42 -g 0 -l 512 |\
grlogsp -l 512 -x 8 -O 2 -c "(b) LPC-MCEP" ;\
\
bcut +f -n 12 -s 10 -e 135 < data.mcep.mcep |\
mgc2sp -m 12 -a 0.42 -g 0 -l 512 |\
grlogsp -l 512 -x 8 -O 3 -c "(c) MCEP" ;\
\
bcut +f -n 12 -s 10 -e 135 < data.mgcep.mcep |\
mgc2sp -m 12 -a 0.42 -g 0 -l 512 |\
grlogsp -l 512 -x 8 -O 4 -c "(d) MGCEP-MCEP" ) | xgr
```



## 8 Playing with the Vocoder Based on Mel-Cepstrum

### 8.1 High- or low-pitched voice

**Files:** [data.mcep.high.syn](#): synthesized speech (float)

[data.mcep.low.syn](#): synthesized speech (float)

```
sopr -m 0.4 data.pitch |\
excite -p 80 | mlsadf -m 20 -a 0.42 -p 80 data.mcep |\
tee data.mcep.high.syn | da +f -s 16
```

```
sopr -m 2 data.pitch |\
excite -p 80 | mlsadf -m 20 -a 0.42 -p 80 data.mcep |\
tee data.mcep.low.syn | da +f -s 16
```



## 8.2 Fast- or slow-speaking voice

**Files:** [data.mcep.fast.syn](#): synthesized speech (float)

[data.mcep.slow.syn](#): synthesized speech (float)

```
sopr -m 1 data.pitch |\
excite -p 40 | mlsadf -m 20 -a 0.42 -p 40 data.mcep |\
tee data.mcep.fast.syn | da +f -s 16
```

```
sopr -m 1 data.pitch |\
excite -p 160 | mlsadf -m 20 -a 0.42 -p 160 data.mcep |\
tee data.mcep.slow.syn | da +f -s 16
```

## 8.3 Hoarse voice

**Files:** [data.mcep.hoarse.syn](#): synthesized speech (float)

```
sopr -m 0 data.pitch |\
excite -p 80 | mlsadf -m 20 -a 0.42 -p 80 data.mcep |\
tee data.mcep.hoarse.syn | da +f -s 16
```

## 8.4 Robotic voice

**Files:** [data.mcep.robot.syn](#): synthesized speech (float)

```
train -p 200 -l -1 | mlsadf -m 20 -a 0.42 -p 80 data.mcep |\
tee data.mcep.robot.syn | da +f -s 16
```

## 8.5 Child-like or deep voice

**Files:** [data.mcep.child.syn](#): synthesized speech (float)

[data.mcep.deep.syn](#): synthesized speech (float)

```
sopr -m 0.4 data.pitch |\
excite -p 80 | mlsadf -m 20 -a 0.1 -p 80 data.mcep |\
tee data.mcep.child.syn | da +f -s 16
```

```
sopr -m 2 data.pitch |\
excite -p 80 | mlsadf -m 20 -a 0.6 -p 80 data.mcep |\
tee data.mcep.deep.syn | da +f -s 16
```

## 8.6 Various voices

**Files:** [data.float](#): original speech (float)

[data.mcep.syn](#): synthesized speech (float)

[data.mcep.{high, low, fast, slow, hoarse, robot, child, deep}.syn](#): synthesized speech (float)

```
da +f -v -s 16 data.float data.mcep.syn \
data.mcep.{high, low, fast, slow, hoarse, robot, child, deep}.syn
```

## 9 Speech Synthesis Based on HMM

### 9.1 Speech parameter generation from a sequence of HMMs

**Files:** sample.pdf: sequence of mean and variance corresponding to a state sequence included in this example (float, little endian)<sup>3</sup>  
sample.mcep: mel-cepstrum generated from a sequence of HMMs (float)

**Conditions:** analysis order: 24  
weight coefficients for calculating delta:  $w(-1) = -0.5, w(0) = 0, w(1) = 0.5$   
weight coefficients for calculating delta-delta:  $w(-1) = 0.25, w(0) = -0.5, w(1) = 0.25$

**Note:** The state sequence is determined according to the state duration densities of the HMMs. The algorithm is not included in SPTK-3.2.

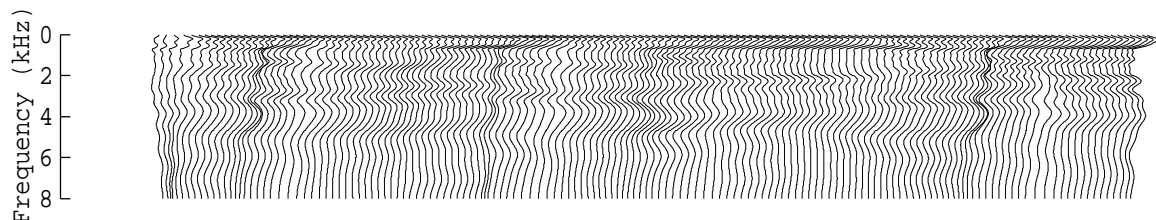
```
mlpg -m 24 -i 1 -d -0.5 0 0.5 -d 0.25 -0.5 0.25 sample.pdf > sample.mcep
```

### 9.2 Plotting spectra calculated from generated mel-cepstrum

**Files:** sample.mcep: mel-cepstral coefficients (float)

**Conditions:** analysis order: 24  
frequency warping parameter:  $\alpha = 0.42$   
FFT size: 512 points  
plotted frames: from 100-th to 250-th  
sampling frequency: 16 kHz

```
bcut +f -n 24 -s 100 -e 250 < sample.mcep |\nmgc2sp -m 24 -a 0.42 -g 0 -l 512 | grlogsp -l 512 -x 8 -t | xgr
```



### 9.3 Speech synthesis from the generated mel-cepstrum

**Files:** sample.pitch: pitch data generated from a sequence of MSD-HMMs included in this example (float, little endian)<sup>4</sup>  
sample.mcep: mel-cepstrum (float)  
[sample.mcep.syn](#): synthesized speech (float)

**Conditions:** frame period: 80 points (5 ms)  
analysis order: 24  
frequency warping parameter:  $\alpha = 0.42$

<sup>3</sup>If you compiled SPTK with "--enable-double" option, please first convert this file into double format:  
`x2x +sd sample.pdf > sample.pdf.double`

<sup>4</sup>If you compiled SPTK with "--enable-double" option, please first convert this file into double format:  
`x2x +sd sample.pitch > sample.pitch.double`



### 9.4.2 Dump variance vectors of static feature vectors

```
bcp +f -l 150 -s 75 -e 99 sample.pdf | sopr -INV | dmp -n 24 | less
```

### 9.4.3 Dump dynamic feature vectors (delta)

```
bcp +f -l 150 -s 25 -e 49 sample.pdf | dmp -n 24 | less
```

### 9.4.4 Dump variance vectors of dynamic feature vectors (delta)

```
bcp +f -l 150 -s 100 -e 124 sample.pdf | sopr -INV | dmp -n 24 | less
```

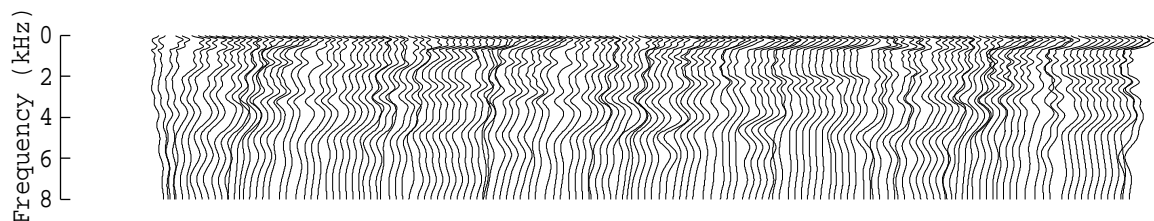
## 9.5 Speech synthesis without dynamic feature

**Files:** sample.pitch: pitch data generated from a sequence of MSD-HMMs (float)  
sample.mcep.wo-dyn: mel-cepstrum generated without dynamic feature (float)  
[sample.mcep.wo-dyn.syn](#): synthesized speech without dynamic feature (float)

**Conditions:** frame period: 80 points (5 ms)  
analysis order: 24  
frequency warping parameter:  $\alpha = 0.42$

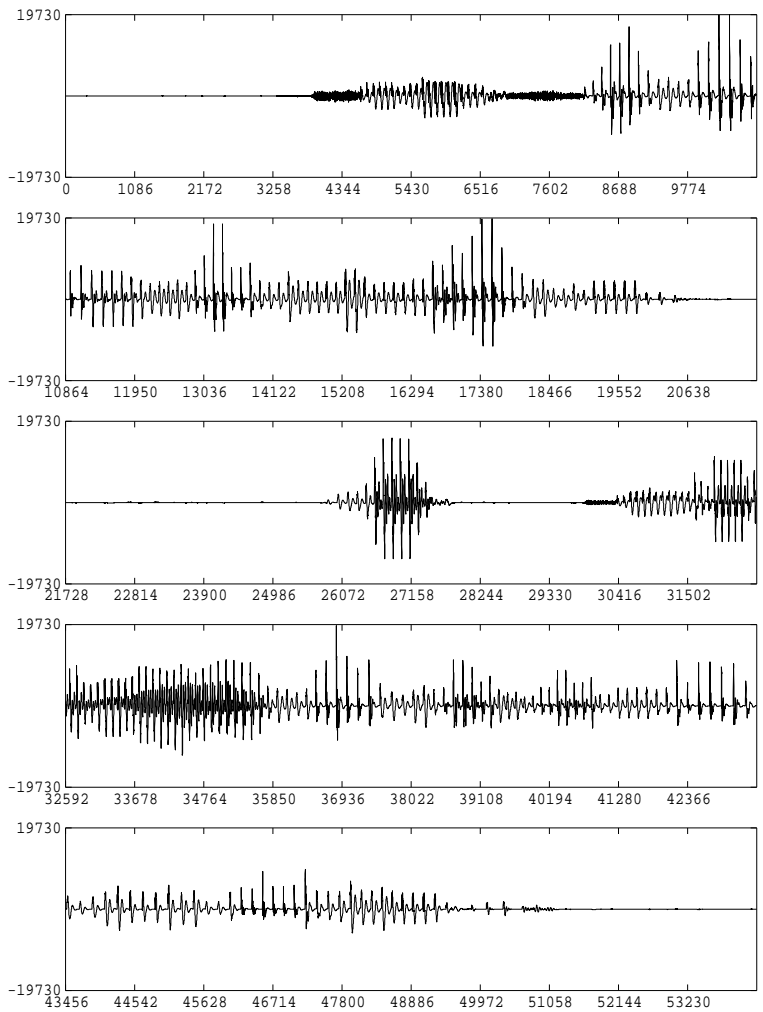
```
bcp +f -l 150 -s 0 -e 24 sample.pdf > sample.mcep.wo-dyn
```

```
bcut +f -n 24 -s 100 -e 250 < sample.mcep.wo-dyn |\nmgc2sp -m 24 -a 0.42 -g 0 -l 512 | grlogsp -l 512 -x 8 -t | xgr
```



```
excite -p 80 sample.pitch |\nmlsadf -p 80 -a 0.42 -m 24 sample.mcep.wo-dyn > sample.mcep.wo-dyn.syn
```

```
gwave +f sample.mcep.wo-dyn.syn | xgr
```



da +f -s 16 sample.mcep.wo-dyn.syn sample.mcep.syn