SEARCHING FOR A ROBUST MFCC-BASED PARAMETERIZATION FOR ASR APPLICATION

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Keywords: MFCC parameterization, critical band-pass filters, robust front-end.

Abstract: The paper concerns with searching for areas of robust setting a MFCC-based parameterization as regards numbers of band-pass filters and computed coefficients. Settings that are theoretically recommended for telephone and microphone speech are compared with a large number of experimental results and a new technique for determination of robust areas of {<# of band-pass filters>×<# of coefficients>} is designed.

1 INTRODUCTION

The state of the art parameterization techniques used in ASR systems try to model the process of human hearing. In speech processing terminology these techniques are known as MFCC (Zheng and Song, 2001) and PLP parameterizations. It is well known that both these techniques attempt to accommodate the parameter estimation process to the way of human hearing and how human perceive sounds with various frequencies. However, one question that we have to deal with is a selection of an "optimal" number of critical band-pass filters and a number of computed coefficients. In papers published in many prestige world conferences we usually find nearly always the same settings without necessary analysis of the task conditions and reference e.g. to the used sampling frequency of speech signal (perhaps it is influenced by the default setting the software tool HTK, which is frequently used at many research labs). On the other hand, from the relatively rich experience of building many ASR systems we known that there isn't only one universal setting which would yield for given "quality" of speech signal the most successful results of recognition experiments. Experimental results however indicate that the best classification results create in the space {<number of band-pass filters> × <number of coefficients>} certain areas in which the successfulness is high and it doesn't change too much (i.e. it doesn't dependent on the change of the number of critical band-filters and the number of coefficients). The goal of described works is to find settings (i.e. the number of filters and derived coefficients), which correspond to the best recognition results and then for such solutions to specify "areas of robust setting".

The whole work is done with the MFCC parameterization and for speech data of telephone ($F_v = 8 \text{ kHz}$) and microphone ($F_v = 44.1 \text{ kHz}$) quality.

2 MFCC BASED PROCESSING

The computational algorithm of the MFCC parameterization is realized by the bank of symmetric overlapping triangular filters spaced linearly in a mel-frequency axis, according to auditory perceptual considerations. The spacing as well as bandwidth of the particular filters is determined by a critical-band concept. To execute this process we have to perform following steps:

•Computation of short-term speech spectrum.

•Non-linear frequency transformation and criticalband spectral resolution – triangular band-pass filters in a mel-frequency axis.

Table 1: Recommended numbers of filters for different values of sampling frequency.

Sampling frequency <i>F</i> _v [kHz]	Band width [kHz]	Band width [mell]	Number of filters M
8	0÷4	0÷2146	15
16	0÷8	0÷2840	20
44.1	0÷22	0÷3921	27

• Computation of cepstral coefficients.

• Applying an inverse discrete Fourier transform.

For the final acoustic modelling we extended the original MFCC representation with derived delta and delta-delta features. See Table 1 for recommended numbers of filters based on a critical-band concept for different values of sampling frequency.

3 SEARCHING FOR ROBUST AREAS

We suggested following approach the to determination of areas of robust parameter settings: Searching for lower boundary of the number of band-pass filters. To find the lower boundary of a robust area, i.e. left from the point of view a minimum number of applied band-pass filters (see Table 2 and 3), we chose such a statistic which calculates for each number of band-pass filters the average of the 5 best recognition results (Acc) obtained for different number of coefficients. Let us define the recognition accuracy for f band-pass filters and c coefficients as $A_{f,c}$. Then to determine the average value of the 5 best recognition results for given number of band-pass filters we have to order firstly results $A_{f,c}$ according to the size, i.e. we define $A_{f,[i]}$, where $A_{f,[1]} \ge A_{f,[2]} \ge \dots$ and then we compute desired statistic as

$$A_{f,\overline{5}} = \frac{1}{5} \sum_{i=1}^{5} A_{f,[i]} .$$
 (1)

Now we find the maximum of A_{min} for $f0 < f_{\text{min}}$, f_{max} , where f_{min} is minimum and $f_{\text{max}}/\tilde{\text{max}}$ is described by the number of band-pass filters, for which measurements were performed, i.e.

$$A_{\text{Max},\bar{5}} = \max_{f} A_{f,\bar{5}} .$$
 (2)

The lower boundary of the robust area (from the point of view applied band-pass filters) we can define so that we determine the first (for increasing number of filters) value of the number of filters f_{Lbou} , for which the value $A_{f,\overline{5}}$ is greater or equal than 99% of $A_{\text{Max},\overline{5}}$, so

$$f_{\text{Lbou}} = \underset{f}{\operatorname{argmin}} A_{f,\overline{5}} \Big|_{A_{f,\overline{5}} \ge 0.99 A_{\text{Max},\overline{5}}}$$
(3)

Determining lower and upper boundaries of a number of coefficients. Considering that the recognition results don't vary too much for increasing number of band-pass filters and a fixed number of used coefficients it is possible to derive the lower and upper boundary of robust area for the whole set of recognition results. A detail analysis of all results (in Table 2 and 3 we could show – owing to limited space – the results of only a small segment of nearly one thousand performed experiments) indicates that the area of the "best" results shifts slightly towards higher number of coefficients. For that reason the robust area was looked for as the interval $\langle f_1, f_u \rangle = \langle f_{Lbou}, f_{Lbou+9} \rangle$; $\langle f_{Lbou+10}, f_{Lbou+19} \rangle$; A block of 10 band-pass filters was chosen so that the resulting area might contain sufficient number of measurements and calculated statistics could be considered to be evidential (Freund,1998). For individual values of a number of coefficients *c*, $c0 < c_{\min}, c_{\max} \rangle$ (where c_{\min} and c_{\max} are respectively values of minimum and maximum number of coefficients for which measurements were performed) we determined average values $\bar{A}_{< l,u>,c}$ (in intervals $\langle f_l, f_u \rangle$)

$$\overline{A}_{,c} = \frac{1}{u-l+1} \sum_{i=l}^{u} A_{i,c}, \qquad c \in < c_{\min}, c_{\max} > \quad (4)$$

Now we can define $\bar{A}_{< l, u>, Max}$ as

$$A_{\langle l,u\rangle,\text{Max}} = \max_{c} A_{\langle l,u\rangle,c} \tag{5}$$

and then to determine the value of a number of coefficients for which this maximum occurred

$$c_{\langle l,u\rangle}^{\text{Max}} = \underset{c}{\operatorname{argmax}} \overline{A}_{\langle l,u\rangle,c} , \qquad (6)$$

where $c0 < c_{\min}$, $c_{\max} >$. Now we can define the lower $c^{L}_{<l,u>}$ and upper $c^{U}_{<l,u>}$ boundary of the robust setting from the point of view a number of coefficients. The desired interval was defined by the values which don't fall below 99% of $\bar{A}_{<l,u>,c}$

$$c_{\langle l,u\rangle}^{L} = \underset{c}{\operatorname{argmin}} \overline{A}_{\langle l,u\rangle,c} \left|_{\overline{A}_{\langle l,u\rangle,c} \ge 0.99\overline{A}_{\langle l,u\rangle,Max}}, \right. (7)$$

$$c_{\langle l,u\rangle}^{U} = \underset{c}{\operatorname{argmax}} \overline{A}_{\langle l,u\rangle,c} \left|_{\overline{A}_{\langle l,u\rangle,c} \ge 0.99\overline{A}_{\langle l,u\rangle,Max}}, \right. (8)$$

For this area we can define the value $f_{<c^{L},c^{U}>}^{Max}$ as the number of filters for which $\overline{A}_{f,<c^{L},c^{U}>}$ attains its maximum (i.e. its "optimum" or rather "recommended" value of a number of band-pass filters) for $<f_{l}, f_{u}>$. Now we can define

$$\overline{A}_{\text{Max}, } = \max_{f} \overline{A}_{f, }, f \in$$
(9)
$$f_{\text{Max}, }^{\text{Max}} = \operatorname*{argmax}_{f} \overline{A}_{f, }, f \in$$
(10)

The area of robust setting. From the above recommendations we can now determine the area of robust setting of the number of band-pass filters and coefficients as

robust area = { $f \in \langle f_l, f_u \rangle \times c \in \langle c_{\langle l,u \rangle}^{L}, c_{\langle l,u \rangle}^{U} \rangle$ } (11) The mean and deviation computed from recognition results in this area give us a measure of quality for given settings.

4 EXPERIMENTAL RESULTS

As was presented above, all experiments were performed using speech data sets of two different qualities: telephone and microphone. The telephone-based corpus consists of Czech read speech transmitted over a telephone channel. One hundred speakers were asked to read various sets of 40 sentences. The microphone-based corpus (highquality speech) is a read-speech database consisting of speech of 100 speakers. Each speaker read a set of 40 sentences (same as in the telephone-based case). The telephone and microphone test sets consisted of 100 sentences randomly selected from utterances of 100 different speakers who were not included in the training databases. The vocabulary in all our test tasks contained 528 different words. There were no OOV words. The basic speech unit of our system is a triphone. Each individual triphone is represented by a three states HMM; each state has 8 mixtures of multivariate Gaussians. In all recognition experiments a language model based on zerograms was applied. For that reason the perplexity of the task was 528.

MFCC parameterization with telephone data

To find areas of robust settings we systematically built and tested nearly one thousand ASR systems. In fact it was for f0<8,45> and c0<4,30>. Recognition results of these experiments are summarized in Table 2 and depicted in Figure 1 (for lack of space Table 2 shows only a part of these results).



Figure 1: MFCC telephone-quality data.

Figure 2 shows the dependency of the average of the 5 best results on the number of band-pass filters. The frequency for which the $\overline{A}_{f,\overline{5}}$ exceeds $0.99 \overline{A}_{\text{Max},\overline{5}}$ is $f_{\text{Lbou}}=14$. In Table 4 you can find all important statistics needed to determine areas of robust settings. It is evident that from the point of view the number of band-pass filters the first area begins by crossing boundary f_{Lbou} . An increasing number of applied band-pass filters above this boundary has practically no influence to the recognition accuracy.



Figure 2: Dependency of $\overline{A}_{f,\overline{5}}$ on the number of filters.

The robust area f0<12, $21>\times c0<10,14>$ and the recommended setting f=15 and c=12 are in a very good agreement with theoretically derived value (M=15) enumerated in Table 1. Also the default HTK setting (i.e. 13 coefficients) can be considered to be correct even though a smaller number coefficients (c0<10, 14>) is also appropriate.

MFCC parameterization with microphone data The area of robust setting for microphone data was searched in fact for f0<18,45> and c0<4,30>. Results of recognition experiments are summarized in Table 3 and depicted in Figure 3.



Figure 3: MFCC microphone-quality data.

Figure 4 shows that for microphone speech the value exceeds $0.99\overline{A}_{Max,\overline{5}}$ for $f_{Lbou}=25$. Similarly as in a case of telephone speech the recognition accuracy changes for increasing number of band-pass filters only slightly. However the area of robust setting is here broader, c0<14,23>.



Figure 4: Dependency of $A_{f,\bar{5}}$ on the number of filters.

Let us note that this interval doesn't contain the HTK default setting, i.e. the value of 13 coefficients. The robust area $f0<22,31>\times c0<14,23>$ and the recommended setting f=29 and c=17 are again in a relatively good agreement with theoretically derived value (M=27) given in Table 1.The mean and

deviation computed from recognition results in this area give us a measure of quality for given settings.

	$f 0 < f_l, f_u >$								
	<i>l</i> =12, <i>u</i> =21	<i>l</i> =22, <i>u</i> =31							
	(telephone)	(microphone)							
$\overline{A}_{< l, u >, \text{Max}}$ [%]	84,60	89,73							
$c_{}^{\text{Max}}$	12	17							
$c_{/}^{\mathrm{L}}c_{}^{\mathrm{U}}$	10 / 14	14 / 23							
$\overline{A}_{\text{Max}, < c^{\text{L}}}, c^{\text{U}} > [\%]$	84,83	89,62							
$f_{\langle c^{\mathrm{L}}, c^{\mathrm{U}} \rangle}^{\mathrm{Max}}$	15	29							
Robust area	f 0<12,21>× × c 0<10, 14>	f0<22,31>× × c0<14,23>							
Recomm. setting	f=15; c=12	f=29; c=17							
# of measures	50	100							
Average of Acc	84,24	89,40							
Deviation of Acc	0,76	0,53							

5 CONCLUSIONS

The MFCC-based parameterization is a very efficient tool for description of speech in ASR systems. We showed that the theory of critical-bands

of hearing is both for telephone ($F_v=8kHz$) and microphone ($F_v=44.1kHz$) speech data in a good agreement with experimental results. Very useful conclusions were obtained for the numbers of "robust" coefficients for which the ASR system demonstrates comparable recognition accuracy.

ACKNOWLEDGEMENTS

This paper was supported by the AVCR, project no. 1QS101470516 and the project of the EU 6th FP no. IST-034434.

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Table 2: Recognition accuracy for various numbers of filters and parameters for telephone data.

Average for <i>f</i> 0<12,21>	# filters # coeff.	10	11	12	13	14	15	16	17	18	19	20	21	22	23
83,12	9	83,98	82,95	83,76	83,10	83,76	83,25	83,32	82,88	82,81	82,95	81,93	83,39	82,15	82,07
84,01	10	84,13	83,47	85,67	84,50	84,86	83,91	83,10	85,01	83,84	83,10	83,54	83,32	82,81	82,51
84,25	11	83,84	83,61	84,42	84,13	84,42	84,20	84,42	84,86	83,76	83,91	84,20	84,20	84,28	83,61
84,60	12	83,17	83,10	83,32	84,79	84,42	85,75	84,35	84,28	85,16	85,01	84,86	84,06	82,95	83,47
84,47	13	81,78	82,22	85,82	83,91	85,08	85,45	84,72	83,98	82,88	83,47	84,57	85,30	84,42	84,42
83,86	14	79,50	81,63	81,85	84,42	84,64	84,86	83,91	83,54	82,59	83,84	84,35	84,64	84,28	83,54
83,46	15	80,53	80,68	81,78	81,56	85,16	84,57	83,32	83,69	82,22	83,84	84,50	83,91	83,76	84,72
Average	of the 5	83,45	83,16	84,34	84,35	84,83	84,91	84,50	84,38	83,69	84,07	84,50	84,42	84,07	83,95

Table 3: Recognition accuracy for various numbers of filters and parameters for microphone data.

Average for $f0 < 22.31 >$	# filters # coeff.	20	21	22	23	24	25	26	27	28	29	30	31	32	33
88,62	13	87,51	89,29	88,15	86,72	87,08	87,79	88,51	89,94	89,65	89,65	89,36	89,36	89,65	89,22
89,34	14	88,72	88,87	88,87	88,65	89,44	90,01	89,08	89,51	89,58	89,44	89,42	89,44	89,52	89,44
89,36	15	89,01	88,94	89,29	88,08	89,01	89,44	89,51	90,22	89,36	89,22	89,65	89,86	89,51	89,72
89,60	16	88,94	89,72	89,72	90,15	89,65	89,08	89,36	89,51	89,36	89,35	90,01	89,79	90,22	89,51
89,73	17	88,87	89,22	90,08	89,58	89,44	90,01	89,58	89,94	89,58	89,88	89,36	89,86	90,29	89,62
89,67	18	88,15	89,08	89,02	89,68	89,36	89,31	89,44	89,88	89,58	90,51	89,94	89,94	89,51	89,72
89,58	19	88,65	88,87	90,08	89,36	89,58	89,36	88,87	89,41	89,86	90,01	89,94	89,36	90,36	89,51
89,43	20	88,72	89,65	89,29	90,08	88,87	90,29	89,86	89,44	89,01	88,29	90,08	89,08	89,79	90,22
89,39	21	88,94	89,58	89,15	88,94	88,87	89,65	89,22	89,35	90,15	90,01	89,35	89,22	90,08	89,58
89,19	22	87,51	86,37	88,44	89,72	89,22	89,58	89,51	88,72	89,22	89,79	88,94	88,79	90,22	89,36
89,40	23	87,08	86,80	87,22	89,86	91,22	89,22	89,79	88,87	89,94	89,72	88,94	89,22	88,94	90,01
88,78	24	85,94	87,15	86,94	89,94	88,37	88,65	89,79	88,72	89,29	89,22	88,65	88,22	88,51	89,72
Average of th	e 5 best	88,9	89,39	89,69	89,95	89,87	89,88	89,78	89,97	89,75	90,04	89,94	89,78	90,23	89,88