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MLSA Digital Filter

Outline

- Introduction
- Approximate Representation of the Mel Scale
 - S domain to Z domain Conversion
 - Z domain to Z domain Conversion
 - Frequency Warping
- Filter with Exponential Transfer function
- Pade Approximation
- Modification of Basic Transfer function
- Realization

Introduction

- Lower order Mel cepstrum coefficients can be used to synthesize speech with satisfactory quality
- The Mel log spectrum is approximated by a recursive filter
- Strube in [1] proposed a recursive filter for nonlinear linear predication
- Strube's filter has high sensitivity and wider distributions
- In this paper another filter is formulated which has better sensitivity and narrow distributions

Approximate Representation of the Mel Scale[2]

- S to Z domain mapping:

$$g(t) = \int_{-\infty}^{+\infty} f(\tau)h(t - \tau)d\tau \triangleq f(t) \otimes h(t) \qquad g_n = \sum_{k=-\infty}^{+\infty} f_k h_{n-k} \triangleq f_n * h_n$$

$$x_n + cy_n \Rightarrow x(t) + cy(t) \qquad f_n = \sum_{k=-\infty}^{+\infty} f_k \delta_{n-k} \Rightarrow f(t) = \sum_{k=-\infty}^{+\infty} f_k \phi_k(t)$$

$$F_L(s) = \sum_{n=-\infty}^{+\infty} f_n \Phi_n(s)$$

Approximate Representation of the Mel Scale[2]

- S to Z domain mapping:

$$G_L(s) = F_L(s)H_L(s).$$

Represents a Continuous Convolution

$$\sum_{n=-\infty}^{+\infty} g_n \Phi_n(s) = \sum_{r=-\infty}^{+\infty} \sum_{k=-\infty}^{+\infty} f_k h_r \Phi_r(s) \Phi_k(s).$$

$$n = r + k$$

$$\sum_{n=-\infty}^{+\infty} \sum_{k=-\infty}^{+\infty} f_k h_{n-k} \Phi_n(s) = \sum_{n=-\infty}^{+\infty} \sum_{k=-\infty}^{+\infty} f_n h_{n-k} \Phi_{n-k}(s) \Phi_k(s).$$

$$\Phi_n(s) = \Phi_{n-k}(s) \Phi_k(s) \quad \Phi_n(s) \Phi_m(s) = \Phi_{n+m}(s)$$

$$\Phi_{n+1}(s) - \Phi_1(s) \Phi_n(s) = 0$$

$$\Phi_n(s) = c[\Phi_1(s)]^n.$$

$$\Phi_n(s) = [\Phi_1(s)]^n.$$

Approximate Representation of the Mel Scale[2]

- S to Z domain mapping:
- If the previous condition is satisfied then $f(t) = \sum_{k=-\infty}^{+\infty} f_k \phi_k(t)$ can be thought of as a mapping from s to z

$$F_D(z) = \sum_{n=-\infty}^{+\infty} f_n z^{-n}.$$

$$F_L(s) = \sum_{n=-\infty}^{+\infty} f_n \{ [\Phi_1(s)]^{-1} \}^{-n}$$

$$z = [\Phi_1(s)]^{-1} \triangleq M(s)$$

Approximate Representation of the Mel Scale[2]

- Z to Z domain mapping:

$$f(t) = \sum_{n=-\infty}^{\infty} f_n \phi_n(t) \quad f(t) = \sum_{k=-\infty}^{\infty} g_k \lambda_k(t)$$

$$\lambda_k(t) = \sum_{n=-\infty}^{\infty} \psi_{k,n} \phi_n(t)$$

$$\sum_{n=-\infty}^{\infty} f_n \phi_n(t) = \sum_{k=-\infty}^{\infty} g_k \lambda_k(t) = \sum_{k=-\infty}^{\infty} g_k \sum_{n=-\infty}^{\infty} \psi_{k,n} \phi_n(t)$$

$$\sum_{n=-\infty}^{\infty} f_n \phi_n(t) = \sum_{n=-\infty}^{\infty} \sum_{k=-\infty}^{\infty} g_k \psi_{k,n} \phi_n(t)$$

$$f_n = \sum_{k=-\infty}^{\infty} g_k \psi_{k,n}$$

Approximate Representation of the Mel Scale[2]

- Z to Z domain mapping:
- The properties required of the set of sequences $\{\psi_k(z)\}$ in order that convolution is preserved can be determined in a manner analogous to s to z mapping

$$\psi_r(z)\psi_{k-r}(z) = \psi_k(z)$$

$$\psi_k(z) = [\psi_1(z)]^k$$

Approximate Representation of the Mel Scale[2]

- Z to Z domain mapping:

$$f_n = \sum_{n=-\infty}^{\infty} g_k \psi_{k,n}$$

$$F(z) = \sum_{k=-\infty}^{\infty} g_k \psi_k(z)$$

$$\sum_{k=-\infty}^{\infty} f_k z^{-k} = \sum_{k=-\infty}^{\infty} g_k \left[\{\psi_1(z)\}^{-1} \right]^{-k}$$

$$Z = \{\psi_1(z)\}^{-1} = m(z)$$

$$F(z) = G(m(z))$$

Thus changing from the representation f_n to g_k is equivalent to mapping the complex variable plane z for f_n to the Z' for g_k by means of a substitution of variables $z'=m(z)$

Frequency Warping[2]

- Letting Ω be angular frequency in the z plane and Ω' be angular frequency in the z' plane
- We want $z'=m(z)$ to satisfy

$$e^{j\Omega} = m[e^{j\Omega'}] \quad \Omega = \theta(\Omega')$$

- To satisfy this and the convolution preservation condition we have

$$\psi_k(e^{j\Omega}) = e^{-jk\theta(\Omega')}$$

Frequency Warping[2]

- Thus the Fourier transform of $\psi_k(n)$ must have an all-pass characteristic
- The negative of the phase will then correspond to the mapping between the frequency axes

$$\Psi_k(z) = \left(\frac{z^{-1} - a^*}{1 - az^{-1}} \right)^k$$

$$\hat{\Omega} = \theta(\Omega) = \arctan \left[\frac{(1 - a^2) \sin \Omega}{(1 + a^2) \cos \Omega - 2a} \right]$$

Mel Cepstrum

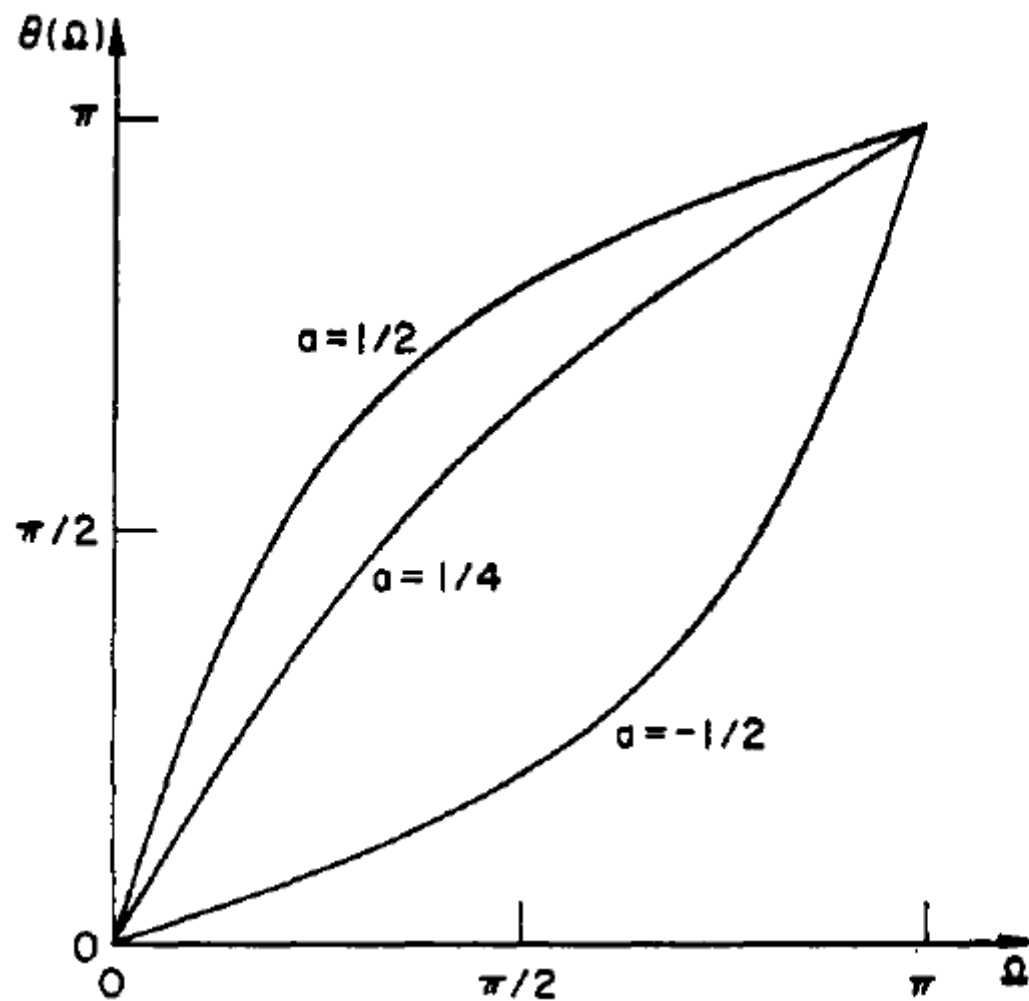
$$G_{\alpha}(\tilde{\Omega}) = G_0(\beta_{\alpha}^{-1}(\tilde{\Omega}))$$

$$\beta_{\alpha}^{-1}(\tilde{\Omega}) = \tan^{-1} \frac{(1 - \alpha^2) \sin \tilde{\Omega}}{(1 + \alpha^2) \cos \tilde{\Omega} + 2\alpha}$$

$$g_{\alpha}(m) = \frac{1}{2\pi} \int_{-\pi}^{\pi} G_{\alpha}(\tilde{\Omega}) e^{jm\tilde{\Omega}} d\tilde{\Omega} = g_{\alpha}(-m)$$

$$c_{\alpha}(m) = \begin{cases} 2g_{\alpha}(m) & (m > 0) \\ g_{\alpha}(m) & (m = 0) \\ 0 & (m < 0) \end{cases}$$

Mel Cepstrum



Filter with Exponential Transfer function[3]

- To approximate the log magnitude response on the mel scale we use a filter with exponential transfer function

$$H_a(\tilde{z}) = \exp (F_a(\tilde{z}))$$

$$F_a(\tilde{z}) = \sum_{m=0}^M c_a(m) \tilde{z}^{-m}$$

$$\ln | H_a(e^{j\tilde{\Omega}}) | = \sum_{m=0}^M c_a(m) \cos (m \tilde{\Omega})$$

Pade Approximation

- The general pade approximation doesn't matches with the formulas given in the papers
- However coding the formulas in software approximates the exponential function
- We will discuss it at the end

Modification of Basic Transfer function[1]

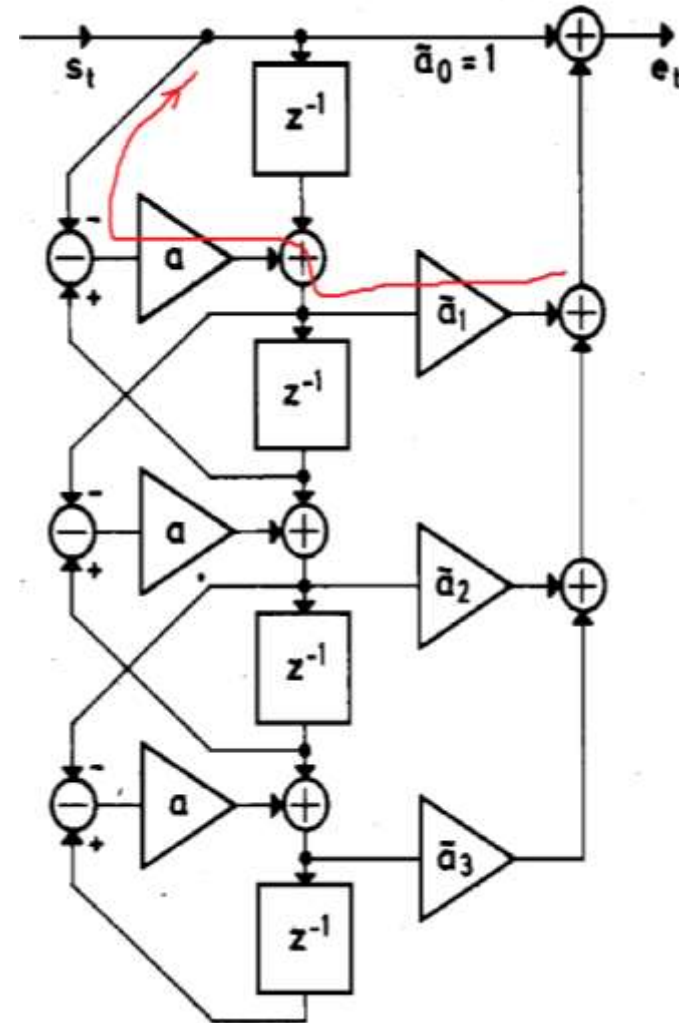
$$\bar{A}(z) = \sum_{k=0}^p \bar{a}_k \bar{z}^{-k}(z), \quad \bar{a}_0 = 1;$$

$$\bar{z}^{-1} = (1 - a^2)z^{-1} / (1 - az^{-1}) - a$$

The all-pass filter is transformed to a lowpass filter

$$\bar{A}(z) = \sum_{k=0}^p b_k z^{-k} (1 - az^{-1})^{-k}.$$

$$b_k = \sum_{n=k}^p C_{kn} \tilde{a}_n, \quad C_{kn} = \binom{n}{k} (1 - a^2)^k (-a)^{n-k}.$$



Modification of Basic Transfer function[1]

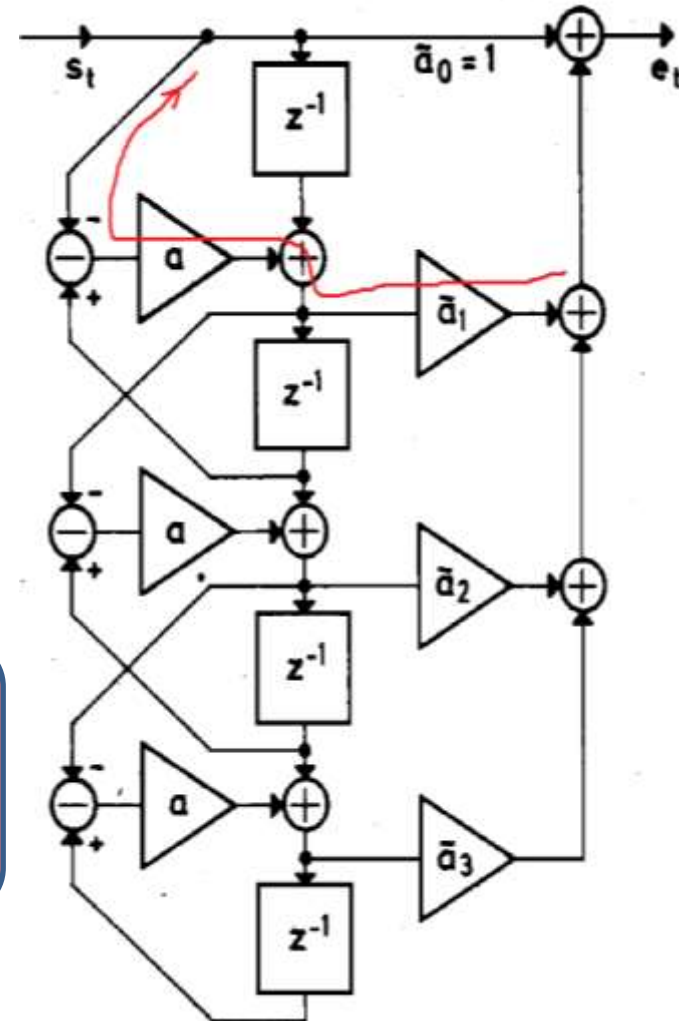
$$\bar{A}(z) = \sum_{k=0}^p \bar{a}_k \bar{z}^{-k}(z), \quad \bar{a}_0 = 1;$$

$$\bar{z}^{-1} = (1 - a^2)z^{-1} / (1 - az^{-1}) - a$$

The all-pass filter is transformed to a lowpass filter

$$\bar{A}(z) = \sum_{k=0}^p b_k z^{-k} (1 - az^{-1})^{-k}.$$

$$b_k = \sum_{n=k}^p C_{kn} \tilde{a}_n, \quad C_{kn} = \binom{n}{k} (1 - a^2)^k (-a)^{n-k}.$$



$$A(z) = \sum_{k=0}^p a_k \tilde{z}^{-k}(z).$$

$$\tilde{z}^{-1} = \frac{(1-a^2)z^{-1}}{(1-az^{-1})} - a.$$

$$A(z) = \sum_{k=0}^p a_k \left[\frac{(1-a^2)z^{-1}}{(1-az^{-1})} - a \right]^k$$

The

$$A(z) = \sum_{k=0}^p z^{-k} (1-az^{-1})^{-k} \left[(1-a^2) - \frac{a(1-az^{-1})}{z^{-1}} \right]^k a_k.$$

$$A(z) = \sum_{k=0}^p b_k z^{-k} (1-az^{-1})^{-k}.$$

$$b_k = \left[(1-a^2) - \frac{a(1-az^{-1})}{z^{-1}} \right]^k a_k.$$

$$b_k = \sum_{n=0}^p$$

$$A(z) = \sum_{k=0}^P a_k \tilde{z}^{-k}(z).$$

$$\tilde{z}^{-1} = \frac{(1-a^2)z^{-1}}{(1-az^{-1})} - a.$$

$$A(z) = \sum_{k=0}^P a_k \left[\frac{(1-a^2)z^{-1}}{(1-az^{-1})} - a \right]^k$$

The

$$A(z) = \sum_{k=0}^P z^{-k} (1-az^{-1})^{-k} \left[(1-a^2) - \frac{a(1-az^{-1})}{z^{-1}} \right]^k a_k.$$

$$A(z) = \sum_{k=0}^P b_k z^{-k} (1-az^{-1})^{-k}.$$

$$b_k = \left[(1-a^2) - \frac{a(1-az^{-1})}{z^{-1}} \right]^k a_k.$$

$$b_k = \sum_{n=0}^P$$

$$\begin{aligned}
\tilde{A}(\tilde{z}) &= \sum_{n=0}^p \tilde{a}_n \tilde{z}^{-n}(z) \\
&= \sum_{n=0}^p \tilde{a}_n \left((1-a^2) z^{-1} / (1-az^{-1}) - a \right)^n \\
&= \sum_{n=0}^p \tilde{a}_n \sum_{k=0}^n \binom{n}{k} (1-a^2)^k z^{-k} (1-az^{-1})^{-k} (-a)^{n-k} \\
&= \sum_{k=0}^p \left[\sum_{n=k}^p \tilde{a}_n \binom{n}{k} (1-a^2)^k (-a)^{n-k} \right] z^{-k} (1-az^{-1})^{-k} \\
&= \sum_{k=0}^p b_k z^{-k} (1-az^{-1})^{-k}
\end{aligned}$$

Modification of Basic Transfer function[3]

- Method II

$$Z^{-1} = \frac{Z^{-1} - \alpha}{1 - \alpha Z^{-1}}$$

$$(1 - \alpha Z^{-1})Z^{-1} = Z^{-1} - \alpha$$

$$Z^{-1} + \alpha = Z^{-1} + \alpha Z^{-1}Z^{-1}$$

$$\alpha + Z^{-1} = Z^{-1}(1 + \alpha Z^{-1})$$

$$Z^{-1} = \frac{(1 + \alpha Z^{-1})}{(\alpha + Z^{-1})}$$

$$F(Z) = \sum_{m=0}^M C(m)Z^{-m}$$

$$(\alpha + Z^{-1}) \sum_{m=0}^M \frac{C(m)}{(\alpha + Z^{-1})} Z^{-m}$$

$$(\alpha + Z^{-1}) \sum_{m=0}^M \frac{C(m)Z^{-1}}{(\alpha + Z^{-1})} Z^{-(m-1)}$$

$$\frac{(1 - \alpha^2)Z^{-1}}{1 - \alpha Z^{-1}} \sum_{m=0}^M B(m)Z^{-(m-1)}$$

$$B(m) = C(m) \left(\frac{Z^{-1}}{\alpha + Z^{-1}} \right)$$

$$B(m+1) = 0$$

$$B(m) = C(m) - \alpha B(m+1)$$

Modification of Basic Transfer function[3]

- Method III

$$F(Z) = \sum_{m=1}^M C(m) Z^{-m}$$

$$F(Z) = \left(\frac{\alpha + Z^{-1}}{1 + \alpha Z^{-1}} \right) \sum_{m=1}^M C(m) \left(\frac{\alpha + Z^{-1}}{1 + \alpha Z^{-1}} \right) Z^{-1} Z^{-(m-1)}$$

$$F(Z) = \left(\frac{\alpha + Z^{-1}}{1 + \alpha Z^{-1}} \right) \sum_{m=1}^M b(m) Z^{-(m-1)}$$

$$F(Z) = z^{-1} \sum_{m=1}^M b(m) Z^{-(m-1)}$$

$$b(m) = c(m) \left(\frac{1 + \alpha Z^{-1}}{\alpha Z + 1} \right)$$

$$\alpha b(m+1) + b(m) = C(m) + \alpha C(m-1)$$

$$b(M+1) = \alpha c(M)$$

$$b(m) = C(m) + \alpha (C(m-1) - b(m+1))$$

$$m = M, M-1, \dots, 2$$

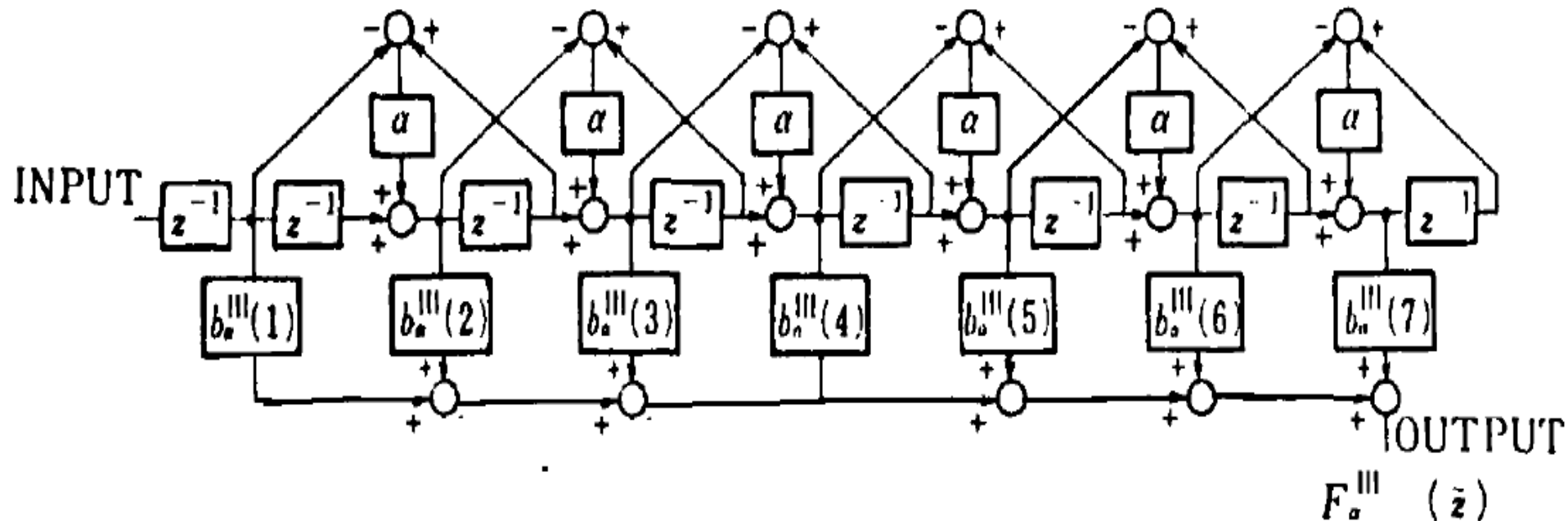
$$b(1) = \frac{C_a(1) - \alpha b(2)}{(1 - \alpha^2)}$$

$$b(0) = C(0) - \alpha b(1)$$

Modification of Basic Transfer function[3]

- Method III

$$b(m) = c(m) \left(\frac{1 + \alpha Z^{-1}}{1 - \alpha Z^{-1}} \right)$$



$$F(Z) = z^{-1} \sum_{m=1}^M b(m) Z^{-(m-1)}$$

Comparison[3]

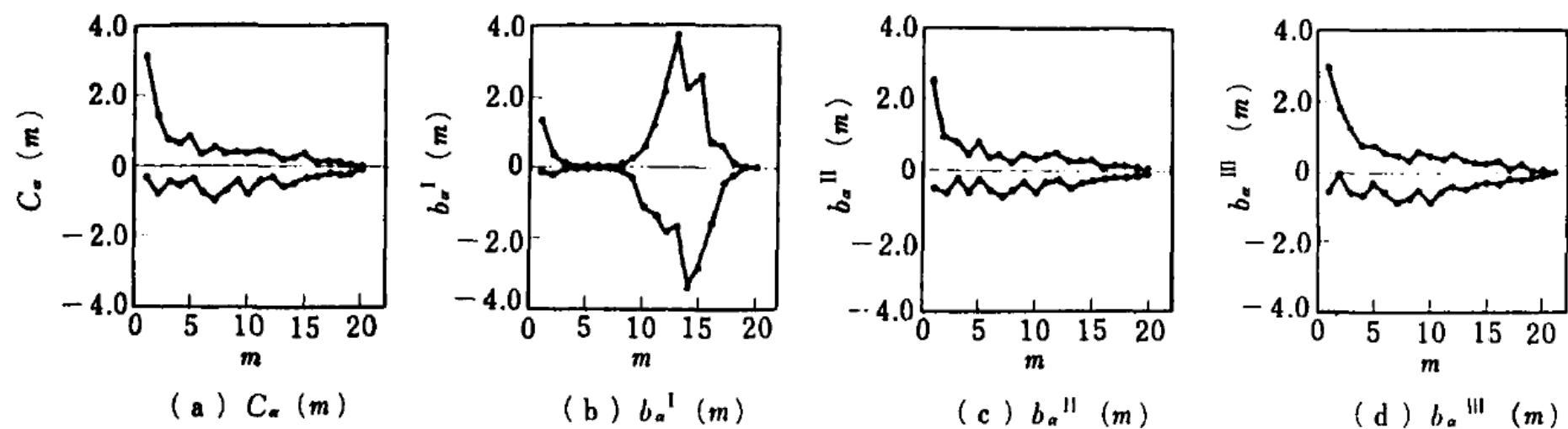


Fig. 1. Maximal and minimal values of mel cepstrum $c_a(m)$ and filter coefficients $b_a^K(m)$ for utterance "nambu dewa higashi no kaze" of a male speaker ($\alpha=0.4$, $K=I, II, III$).

Implementation[3]

- To apply Pade approximation in the range of good approximation, the transfer function of the basic filter is partitioned into the sum of $b(0)$ and $F^n(z)$ ($n = 1 . 2 \dots N$) as

$$F_a^{\text{III}}(\tilde{z}) = b_a^{\text{III}}(0) + \sum_{n=1}^N F_a^{\text{III}, n}(\tilde{z})$$

MLSA filter is given by

$$H_a(\tilde{z}) = \exp (b_a^{\text{III}}(0)) \prod_{n=1}^N R_{L(n)}^{\text{III}} (F_a^{\text{III}, n}(\tilde{z}))$$

Implementation[3]

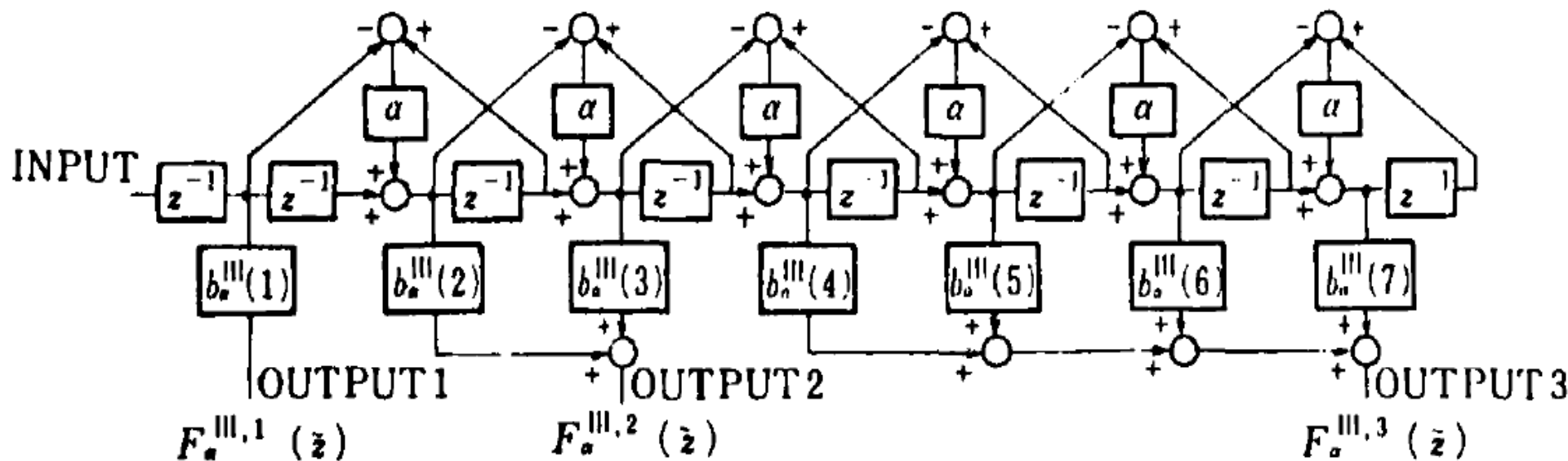
$$F_a^{\text{III}, 1}(\tilde{z}) = z^{-1} b_a^{\text{III}}(1)$$

$$F_a^{\text{III}, 2}(\tilde{z}) = z^{-1} (b_a^{\text{III}}(2) \tilde{z}^{-1} + b_a^{\text{III}}(3) \tilde{z}^{-2})$$

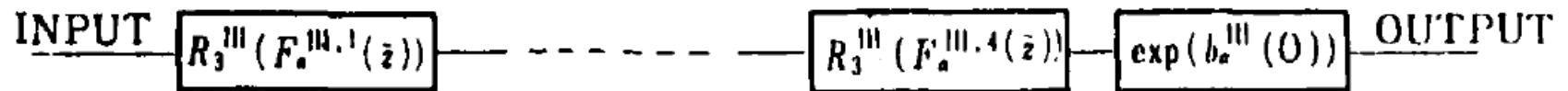
$$F_a^{\text{III}, 3}(\tilde{z}) = z^{-1} (b_a^{\text{III}}(4) \tilde{z}^{-3} + b_a^{\text{III}}(5) \tilde{z}^{-4} \\ + b_a^{\text{III}}(6) \tilde{z}^{-5} + b_a^{\text{III}}(7) \tilde{z}^{-6})$$

$$F_a^{\text{III}, 4}(\tilde{z}) = z^{-1} (b_a^{\text{III}}(8) \tilde{z}^{-7} + b_a^{\text{III}}(9) \tilde{z}^{-8} + \dots \\ \dots + b_a^{\text{III}}(M+1) \tilde{z}^{-M}) \quad (M \leq 20)$$

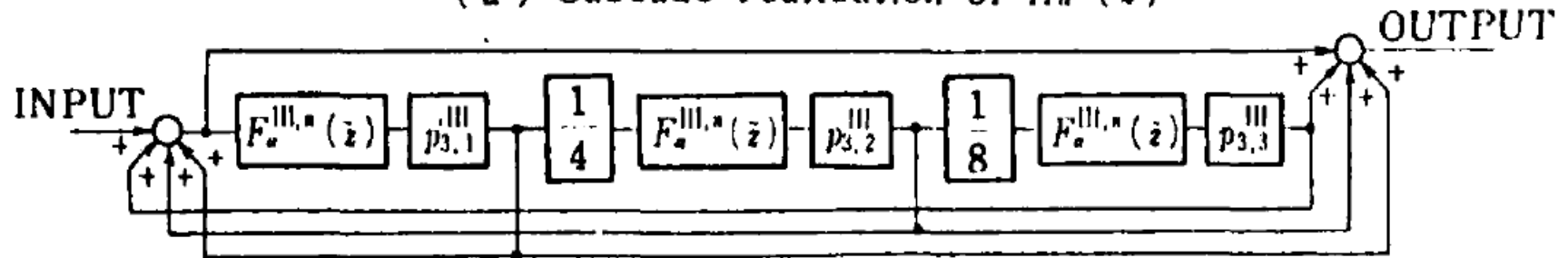
Implementation[3]


$$(c) F_{\sigma}^{III, n}(\tilde{z}) \quad (n=1, 2, 3)$$

Implementation[3]



(a) Cascade realization of $H_a(\bar{z})$



(b) $R_3^{III}(F_a^{III,n}(\bar{z}))$ ($n=1, 2, 3, 4$)

References

1. Strube, Hans Werner. "Linear prediction on a warped frequency scale." The Journal of the Acoustical Society of America 68.4 (1980): 1071-1076.
2. Oppenheim, Alan V., and Donald H. Johnson. "Discrete representation of signals." Proceedings of the IEEE 60.6 (1972): 681-691.
3. Imai, Satoshi, Kazuo Sumita, and Chieko Furuichi. "Mel log spectrum approximation (MLSA) filter for speech synthesis." Electronics and Communications in Japan (Part I: Communications) 66.2 (1983): 10-18.