# ROUNDOFF NOISE MINIMIZATION IN TWO-DIMENSIONAL STATE-SPACE DIGITAL FILTERS USING ERROR FEEDBACK

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# ABSTRACT

This paper considers the problem of minimizing roundoff noise in two-dimensional (2-D) state-space digital filters subject to  $L_2$ -norm dynamic-range scaling constraints. The minimization will be achieved by using error feedback. Several techniques for the determination of the optimal full-scale, block-diagonal, diagonal, and scalar error-feedback matrices for a given 2-D statespace digital filter are proposed. A numerical example is presented to illustrate the utility of the proposed techniques.

## I. INTRODUCTION

One of the primary finite-word-length (FWL) register effects in fixed-point digital filters is the roundoff noise caused by the rounding of products/summations within the realization. One can reduce the roundoff noise at the filter output using error feedback, which is achieved by extracting the quantization error after multiplications and additions, and then feeding the error signal back to a certain point through a simple circuit. Several techniques for error feedback have been presented in the past for 1-D digital filters [1]-[5], and more recently for 2-D digital filters [6]-[9].

This paper proposes several new algorithms for the reduction of roundoff noise in 2-D state-space digital filters. Several closed-form formulas for evaluating the optimal full-scale, block-diagonal, diagonal, and scalar error-feedback matrices for a given 2-D state-space digital filter are derived. A numerical example is presented to illustrate the algorithms proposed and to demonstrate their performance.

## II. 2-D STATE-SPACE DIGITAL FILTERS WITH ERROR FEEDBACK

Consider the Roesser local state-space (LSS) model  $(\mathbf{A}, \mathbf{b}, \mathbf{c}, d)_{m,n}$  which is stable, separately locally con-

trollable and separately locally observable:

$$\begin{aligned} \boldsymbol{x}_{11}(i,j) &= \boldsymbol{A}\boldsymbol{x}(i,j) + \boldsymbol{b}\boldsymbol{u}(i,j) \\ y(i,j) &= \boldsymbol{c}\boldsymbol{x}(i,j) + d\boldsymbol{u}(i,j) \end{aligned} \tag{1}$$

where

Here,  $\boldsymbol{x}^{h}(i,j)$  is an  $m \times 1$  horizontal state vector,  $\boldsymbol{x}^{v}(i,j)$  is an  $n \times 1$  vertical state vector, u(i,j) is a scalar input, y(i,j) is a scalar output, and  $\boldsymbol{A}_{1}$ ,  $\boldsymbol{A}_{2}$ ,  $\boldsymbol{A}_{3}$ ,  $\boldsymbol{A}_{4}$ ,  $\boldsymbol{b}_{1}$ ,  $\boldsymbol{b}_{2}$ ,  $\boldsymbol{c}_{1}$ ,  $\boldsymbol{c}_{2}$ , and d are real constant matrices of appropriate dimensions.

Carrying out the quantization before matrix-vector multiplication, an FWL implementation of (1) can be expressed as

$$\tilde{\boldsymbol{x}}_{11}(i,j) = \boldsymbol{A}\boldsymbol{Q}[\tilde{\boldsymbol{x}}(i,j)] + \boldsymbol{b}\boldsymbol{u}(i,j)$$
  
$$\tilde{\boldsymbol{y}}(i,j) = \boldsymbol{c}\boldsymbol{Q}[\tilde{\boldsymbol{x}}(i,j)] + d\boldsymbol{u}(i,j)$$
(2)

where each component of A, b, c, and d assumes an exact fractional  $B_c$  bit representation. The FWL local state vector  $\tilde{x}(i, j)$  and the output  $\tilde{y}(i, j)$  all have a B bit fractional representation, while the input u(i, j) is a  $(B - B_c)$  bit fraction.

The quantizer  $Q[\cdot]$  in (2) rounds the *B* bit fraction  $\tilde{x}(i, j)$  to  $(B - B_c)$  bits after multiplications and additions, where the sign bit is not counted. The quantization error

$$\boldsymbol{e}(i,j) = \tilde{\boldsymbol{x}}(i,j) - \boldsymbol{Q}[\tilde{\boldsymbol{x}}(i,j)]$$
(3)

coincides with the residue left in the lower part of  $\tilde{\boldsymbol{x}}(i,j)$ . The roundoff error  $\boldsymbol{e}(i,j)$  is modeled as a zeromean noise process of covariance  $\sigma^2 \boldsymbol{I}_{m+n}$  with

$$\sigma^2 = \frac{1}{12} 2^{-2(B-B_c)}.$$

To reduce the filter's roundoff noise, the quantization error e(i, j) is fed back to each input of delay operators through an  $(m + n) \times (m + n)$  constant matrix D in the FWL filter (2). The 2-D filter with error feedback can be characterized by

$$\tilde{\boldsymbol{x}}_{11}(i,j) = \boldsymbol{A}\boldsymbol{Q}[\tilde{\boldsymbol{x}}(i,j)] + \boldsymbol{b}\boldsymbol{u}(i,j) + \boldsymbol{D}\boldsymbol{e}(i,j)$$
  
$$\tilde{\boldsymbol{y}}(i,j) = \boldsymbol{c}\boldsymbol{Q}[\tilde{\boldsymbol{x}}(i,j)] + d\boldsymbol{u}(i,j)$$
(4)

where D is referred to as an *error-feedback matrix*. Subtracting (4) from (1) yields

$$\Delta \boldsymbol{x}_{11}(i,j) = \boldsymbol{A} \Delta \boldsymbol{x}(i,j) + (\boldsymbol{A} - \boldsymbol{D})\boldsymbol{e}(i,j)$$
  
$$\Delta \boldsymbol{y}(i,j) = \boldsymbol{c} \Delta \boldsymbol{x}(i,j) + \boldsymbol{c} \boldsymbol{e}(i,j)$$
(5)

where

$$\Delta \boldsymbol{x}(i,j) = \boldsymbol{x}(i,j) - \tilde{\boldsymbol{x}}(i,j)$$
$$\Delta \boldsymbol{x}_{11}(i,j) = \boldsymbol{x}_{11}(i,j) - \tilde{\boldsymbol{x}}_{11}(i,j)$$
$$\Delta y(i,j) = y(i,j) - \tilde{y}(i,j).$$

Let  $G_D(z_1, z_2)$  be the 2-D transfer function from the quantization error, e(i, j), to the filter output,  $\Delta y(i, j)$ . Then, we obtain

$$G_D(z_1, z_2) = c(Z - A)^{-1}(A - D) + c$$
 (6)

where  $\mathbf{Z} = z_1 \mathbf{I}_m \oplus z_2 \mathbf{I}_n$ . The noise variance gain  $I(\mathbf{D}) = \sigma_{out}^2 / \sigma^2$  is then defined by

$$I(\boldsymbol{D}) = tr[\boldsymbol{W}_D] \tag{7}$$

where  $\sigma_{out}^2$  denotes noise variance at the output, and

$$\boldsymbol{W}_{D} = \frac{1}{(2\pi j)^{2}} \oint_{\Gamma_{1}} \oint_{\Gamma_{2}} \boldsymbol{G}_{D}^{*}(z_{1}, z_{2}) \boldsymbol{G}_{D}(z_{1}, z_{2}) \frac{dz_{1} dz_{2}}{z_{1} z_{2}}$$

with  $\Gamma_i = \{z_i : |z_i| = 1\}$  for i = 1, 2. By applying the 2-D Cauchy integral theorem, we obtain

$$\boldsymbol{W}_D = (\boldsymbol{A} - \boldsymbol{D})^T \boldsymbol{W}_o(\boldsymbol{A} - \boldsymbol{D}) + \boldsymbol{c}^T \boldsymbol{c}$$
 (8)

where  $\boldsymbol{W}_{o}$  is called the *local observability Gramian* of the 2-D filter, and is defined by

$$W_{o} = \frac{1}{(2\pi j)^{2}} \oint_{\Gamma_{1}} \oint_{\Gamma_{2}} (Z^{*} - A^{T})^{-1} c^{T} c (Z - A)^{-1}$$
$$\cdot \frac{dz_{1} dz_{2}}{z_{1} z_{2}} = \sum_{i=0}^{\infty} \sum_{j=0}^{\infty} g(i, j)^{T} g(i, j).$$
(9)

If there is no error feedback in the 2-D filter, then the noise variance gain I(D) with D = 0 becomes

$$I(\mathbf{0}) = \operatorname{tr}[\mathbf{A}^T \mathbf{W}_o \mathbf{A} + \mathbf{c}^T \mathbf{c}]$$
  
= tr[\mathbf{W}\_o]. (10)

The  $l_2$ -norm dynamic-range scaling constraints on the local state vector involves the *local controllability Gramian* of the 2-D filter, which is defined by

$$\boldsymbol{K}_{c} = \frac{1}{(2\pi j)^{2}} \oint_{\Gamma_{1}} \oint_{\Gamma_{2}} (\boldsymbol{Z} - \boldsymbol{A}^{T})^{-1} \boldsymbol{b} \boldsymbol{b}^{T} (\boldsymbol{Z}^{*} - \boldsymbol{A}^{T})^{-1} \cdot \frac{dz_{1} dz_{2}}{z_{1} z_{2}} = \sum_{i=0}^{\infty} \sum_{j=0}^{\infty} \boldsymbol{f}(i, j) \boldsymbol{f}(i, j)^{T}.$$
(11)

The problem considered is to design the errorfeedback matrix D that minimizes (7), where matrix  $W_D$  is specified by (8), subject to that all the diagonal elements of  $K_c$  equal unity.

### III. DETERMINATION OF OPTIMAL ERROR FEEDBACK MATRICES

In this section, we derive closed-form formulas for the determination of the optimal full-scale, blockdiagonal, diagonal, and scalar error-feedback matrix  $\boldsymbol{D}$ to minimize  $I(\boldsymbol{D}) = \text{tr}[\boldsymbol{W}_D]$  for a given 2-D state-space digital filter.

#### Case 1: D is a general matrix

Substituting (8) into (7), we obtain

$$I(\boldsymbol{D}) = \operatorname{tr}[\boldsymbol{c}^{T}\boldsymbol{c} + (\boldsymbol{A} - \boldsymbol{D})^{T}\boldsymbol{W}_{o}(\boldsymbol{A} - \boldsymbol{D})]$$
  
=  $\operatorname{tr}[\boldsymbol{W}_{o}] + \operatorname{tr}[\boldsymbol{D}^{T}\boldsymbol{W}_{o}\boldsymbol{D}] - 2\operatorname{tr}[\boldsymbol{D}^{T}\boldsymbol{W}_{o}\boldsymbol{A}].$ 
(12)

Differentiating (12) with respect to the error-feedback matrix  $\boldsymbol{D}$  yields

$$\frac{\partial I(\boldsymbol{D})}{\partial \boldsymbol{D}} = 2\boldsymbol{W}_o(\boldsymbol{D} - \boldsymbol{A}). \tag{13}$$

By choosing the error-feedback matrix as D = A, the noise gain I(D) in (12) achieves its minimum value

$$I_{min}(\boldsymbol{D}) = \operatorname{tr}[\boldsymbol{W}_o] - \operatorname{tr}[\boldsymbol{A}^T \boldsymbol{W}_o \boldsymbol{A}]$$
  
=  $\operatorname{tr}[\boldsymbol{c}^T \boldsymbol{c}].$  (14)

Case 2: D is a block-diagonal matrix

In this case, matrix D assumes the form

$$\boldsymbol{D} = \boldsymbol{D}_1 \oplus \boldsymbol{D}_4 \tag{15}$$

where  $D_1$  and  $D_1$  are  $m \times m$  and  $n \times n$  matrices, respectively, which leads (12) to

$$I(\boldsymbol{D}) = \operatorname{tr}[\boldsymbol{W}_{o}] + \operatorname{tr}[\boldsymbol{D}_{1}^{T}\boldsymbol{W}_{o1}\boldsymbol{D}_{1}] + \operatorname{tr}[\boldsymbol{D}_{4}^{T}\boldsymbol{W}_{o4}\boldsymbol{D}_{4}] -2\operatorname{tr}[\boldsymbol{D}_{1}^{T}(\boldsymbol{W}_{o1}\boldsymbol{A}_{1} + \boldsymbol{W}_{o2}\boldsymbol{A}_{3})] -2\operatorname{tr}[\boldsymbol{D}_{4}^{T}(\boldsymbol{W}_{o3}\boldsymbol{A}_{2} + \boldsymbol{W}_{o4}\boldsymbol{A}_{4})]$$
(16)

where

$$oldsymbol{W}_o = \left[ egin{array}{ccc} oldsymbol{W}_{o1} & oldsymbol{W}_{o2} \ oldsymbol{W}_{o3} & oldsymbol{W}_{o4} \end{array} 
ight].$$

Letting  $\partial I(\boldsymbol{D})/\partial \boldsymbol{D}_i = \mathbf{0}$  for i = 1, 4 yields

$$D_{1} = A_{1} + W_{o1}^{-1} W_{o2} A_{3}$$
  

$$D_{4} = A_{4} + W_{o4}^{-1} W_{o3} A_{2}.$$
(17)

By substituting (17) into (16), we obtain the minimum value of the noise variance gain  $I(\mathbf{D})$  as

$$I_{min}(\boldsymbol{D}) = \operatorname{tr}[\boldsymbol{W}_o] - tr[\boldsymbol{D}_1^T(\boldsymbol{W}_{o1}\boldsymbol{A}_1 + \boldsymbol{W}_{o2}\boldsymbol{A}_3)] - tr[\boldsymbol{D}_4^T(\boldsymbol{W}_{o3}\boldsymbol{A}_2 + \boldsymbol{W}_{o4}\boldsymbol{A}_4)].$$
(18)

Case 3: D is a diagonal matrix

In this case, matrix D assumes the form

$$D_1 = \operatorname{diag}\{\alpha_1, \alpha_2, \cdots, \alpha_m\}$$
  
$$D_4 = \operatorname{diag}\{\beta_1, \beta_2, \cdots, \beta_n\}$$
(19)

which leads (16) to

$$I(\boldsymbol{D}) = \operatorname{tr}[\boldsymbol{W}_{o}] + \operatorname{tr}[\boldsymbol{D}_{1}^{2}\boldsymbol{W}_{o1}] + \operatorname{tr}[\boldsymbol{D}_{4}^{2}\boldsymbol{W}_{o4}]$$
$$-2\operatorname{tr}[\boldsymbol{D}_{1}(\boldsymbol{W}_{o1}\boldsymbol{A}_{1} + \boldsymbol{W}_{o2}\boldsymbol{A}_{3})] \qquad (20)$$
$$-2\operatorname{tr}[\boldsymbol{D}_{4}(\boldsymbol{W}_{o3}\boldsymbol{A}_{2} + \boldsymbol{W}_{o4}\boldsymbol{A}_{4})].$$

This implies that if  $\alpha_i$ 's and  $\beta_i$ 's satisfy

$$\alpha_{i}\left(\alpha_{i}-2\frac{(\boldsymbol{W}_{o1}\boldsymbol{A}_{1}+\boldsymbol{W}_{o2}\boldsymbol{A}_{3})_{ii}}{(\boldsymbol{W}_{o1})_{ii}}\right)<0,\ i=1,2,\cdots,m$$
$$\beta_{i}\left(\beta_{i}-2\frac{(\boldsymbol{W}_{o3}\boldsymbol{A}_{2}+\boldsymbol{W}_{o4}\boldsymbol{A}_{4})_{ii}}{(\boldsymbol{W}_{o4})_{ii}}\right)<0,\ i=1,2,\cdots,n$$
(21)

then  $I(\mathbf{D}) = \operatorname{tr}[\mathbf{W}_D] < \operatorname{tr}[\mathbf{W}_o]$ . Letting  $\partial I(\mathbf{D}) / \partial \alpha_i = 0$  for  $i = 1, 2, \dots, m$  and letting  $\partial I(\mathbf{D}) / \partial \beta_i = 0$  for  $i = 1, 2, \dots, n$  gives

$$\alpha_{i} = \frac{(\boldsymbol{W}_{o1}\boldsymbol{A}_{1} + \boldsymbol{W}_{o2}\boldsymbol{A}_{3})_{ii}}{(\boldsymbol{W}_{o1})_{ii}}, \ i = 1, 2, \cdots, m$$

$$\beta_{i} = \frac{(\boldsymbol{W}_{o3}\boldsymbol{A}_{2} + \boldsymbol{W}_{o4}\boldsymbol{A}_{4})_{ii}}{(\boldsymbol{W}_{o4})_{ii}}, \ i = 1, 2, \cdots, n$$
(22)

at which  $I(\mathbf{D})$  achieves its minimum as

$$I_{min}(\mathbf{D}) = \operatorname{tr}[\mathbf{W}_{o}] - \sum_{i=1}^{m} \frac{(\mathbf{W}_{o1}\mathbf{A}_{1} + \mathbf{W}_{o2}\mathbf{A}_{3})_{ii}^{2}}{(\mathbf{W}_{o1})_{ii}} - \sum_{i=1}^{n} \frac{(\mathbf{W}_{o3}\mathbf{A}_{2} + \mathbf{W}_{o4}\mathbf{A}_{4})_{ii}^{2}}{(\mathbf{W}_{o4})_{ii}}$$
(23)

where  $(\mathbf{A})_{ii}$  denotes the *i*th diagonal element of a square matrix  $\mathbf{A}$ .

Case 4:  $D_1$  and  $D_4$  are scalar matrices  $\alpha I_m$  and  $\beta I_n$ 

If  $D_1 = \alpha I_m$  and  $D_4 = \beta I_n$  with scalars  $\alpha$  and  $\beta$ , then (20) becomes

$$I(\boldsymbol{D}) = \operatorname{tr}[\boldsymbol{W}_{o}] + \operatorname{tr}[\boldsymbol{W}_{o1}]\alpha^{2} + \operatorname{tr}[\boldsymbol{W}_{o4}]\beta^{2}$$
$$-2\operatorname{tr}[\boldsymbol{W}_{o1}\boldsymbol{A}_{1} + \boldsymbol{W}_{o2}\boldsymbol{A}_{3}]\alpha \qquad (24)$$
$$-2\operatorname{tr}[\boldsymbol{W}_{o3}\boldsymbol{A}_{2} + \boldsymbol{W}_{o4}\boldsymbol{A}_{4}]\beta.$$

Hence, if  $\alpha$  and  $\beta$  satisfy

$$\alpha \left( \alpha - 2 \frac{\operatorname{tr}[\boldsymbol{W}_{o1}\boldsymbol{A}_{1} + \boldsymbol{W}_{o2}\boldsymbol{A}_{3}]}{\operatorname{tr}[\boldsymbol{W}_{o1}]} \right) < 0$$

$$\beta \left( \beta - 2 \frac{\operatorname{tr}[\boldsymbol{W}_{o3}\boldsymbol{A}_{2} + \boldsymbol{W}_{o4}\boldsymbol{A}_{4}]}{\operatorname{tr}[\boldsymbol{W}_{o4}]} \right) < 0$$

$$(25)$$

then  $I(\mathbf{D}) = \text{tr}[\mathbf{W}_D] < \text{tr}[\mathbf{W}_o]$ . Moreover, from  $\partial I(\mathbf{D})/\partial \alpha = 0$  and  $\partial I(\mathbf{D})/\partial \beta = 0$ , it follows that

$$\alpha = \frac{\operatorname{tr}[\boldsymbol{W}_{o1}\boldsymbol{A}_{1} + \boldsymbol{W}_{o2}\boldsymbol{A}_{3}]}{\operatorname{tr}[\boldsymbol{W}_{o1}]}$$

$$\beta = \frac{\operatorname{tr}[\boldsymbol{W}_{o3}\boldsymbol{A}_{2} + \boldsymbol{W}_{o4}\boldsymbol{A}_{4}]}{\operatorname{tr}[\boldsymbol{W}_{o4}]}$$
(26)

which lead (24) to

$$I_{min}(\mathbf{D}) = \text{tr}[\mathbf{W}_{o}] - \frac{(\text{tr}[\mathbf{W}_{o1}\mathbf{A}_{1} + \mathbf{W}_{o2}\mathbf{A}_{3}])^{2}}{\text{tr}[\mathbf{W}_{o1}]} - \frac{(\text{tr}[\mathbf{W}_{o3}\mathbf{A}_{2} + \mathbf{W}_{o4}\mathbf{A}_{4}])^{2}}{\text{tr}[\mathbf{W}_{o4}]}.$$
 (27)

### **IV. A NUMERICAL EXAMPLE**

Let a 2-D state-space digital filter  $(\mathbf{A}, \mathbf{b}, \mathbf{c}, d)_{3,3}$ with d = 0.0 be described by

$$\begin{split} \mathbf{A}_{1} &= \begin{bmatrix} 0.621553 & 0.014666 & -0.476979 \\ -0.081625 & 0.621548 & -0.181986 \\ 0.181983 & 0.476990 & 0.663600 \end{bmatrix} \\ \mathbf{A}_{2} &= \begin{bmatrix} 0.059369 & -0.004829 & -0.024002 \\ -0.646852 & 0.061969 & 0.227715 \\ -0.229635 & 0.021958 & 0.076674 \end{bmatrix} \\ \mathbf{A}_{3} &= \begin{bmatrix} 0.000378 & 0.000763 & 0.001503 \\ -0.000463 & -0.001501 & 0.000812 \\ -0.000021 & -0.000219 & 0.000908 \end{bmatrix} \\ \mathbf{A}_{4} &= \begin{bmatrix} 0.620418 & 0.016504 & -0.479313 \\ -0.083124 & 0.620420 & -0.181961 \\ 0.181967 & 0.479315 & 0.661692 \end{bmatrix} \\ \mathbf{b}_{1} &= \begin{bmatrix} -0.007708 & 0.081835 & 0.028969 \end{bmatrix}^{T} \\ \mathbf{b}_{2} &= \begin{bmatrix} -0.079883 & 0.846271 & 0.294745 \end{bmatrix}^{T} \\ \mathbf{c}_{1} &= \begin{bmatrix} -0.766526 & 0.072050 & 0.267706 \\ 0.07031 & 0.026238 \end{bmatrix}. \end{split}$$

which is stable, separately locally controllable and separately locally observable. This corresponds to the *optimal realization* with minimum roundoff noise  $I(\mathbf{0}) = 4.927082$ , subject to the  $l_2$ -norm dynamic-range scaling constraints.

In case D is allowed to be a general matrix, then (13) suggests that we should choose D = A which yields  $I_{min}(D) = 0.670643$ . Suppose the elements of matrix D are rounded to power-of-two quantization with 3 bits after binary point (integer quantization), then the noise gain is given by

 $I(\mathbf{D}) = 0.700468$  (1.726719).

If D is constrained to be a block-diagonal matrix, then the optimal  $D = D_1 \oplus D_4$  is calculated using (17), which gives

$$\boldsymbol{D}_{1} = \begin{bmatrix} 0.621550 & 0.014658 & -0.476971 \\ -0.081619 & 0.621565 & -0.181992 \\ 0.181976 & 0.476968 & 0.663613 \end{bmatrix}$$
$$\boldsymbol{D}_{4} = \begin{bmatrix} 0.618351 & 0.016703 & -0.478595 \\ -0.082737 & 0.620382 & -0.182106 \\ 0.181250 & 0.479384 & 0.661937 \end{bmatrix}$$

 $I_{min}(D_1 \oplus D_4) = 1.331653.$ 

After 3-bit quantization (integer quantization), this block-diagonal matrix  $D = D_1 \oplus D_4$  gives

 $I(\mathbf{D}_1 \oplus \mathbf{D}_4) = 1.351427$  (2.247670).

If D is constrained to be a diagonal error-feedback matrix, then it can be calculated using (22) as

 $D_1 = \text{diag}\{0.523405, 0.934410, 0.859620\}$ 

 $D_4 = \text{diag}\{0.521174, 0.934021, 0.858810\}$ 

which yields  $I_{min}(\mathbf{D}) = 1.833208$ . After 3-bit quantization (integer quantization), this diagonal matrix  $\mathbf{D} = \mathbf{D}_1 \oplus \mathbf{D}_4$  yields

 $I(\mathbf{D}_1 \oplus \mathbf{D}_4) = 1.840195$  (2.247670).

If a scalar error-feedback matrix is calculated using (26), then we obtain

$$\alpha = 0.772479, \qquad \beta = 0.771335$$

which yields  $I_{min}(\mathbf{D}) = 1.991329$ . After 3-bit quantization (integer quantization), this scalar matrix  $\mathbf{D}$  results in

 $I(\mathbf{D}) = 1.993695$  (2.247670).

From these results, it is observed that the utilization of an optimal error feedback matrix leads to considerable reduction in roundoff noise, even when a scalar  $\boldsymbol{D} = \alpha \boldsymbol{I}_m \oplus \beta \boldsymbol{I}_n$  with quantized  $\alpha$  and  $\beta$ .

#### V. CONCLUSION

In this paper, the problem of minimizing roundoff noise in 2-D state-space digital filters has been investigated by means of error feedback. General, blockdiagonal, diagonal, and scalar error-feedback matrices for minimizing the noise variance gain in a given 2-D state-space digital filter have been derived. Simulation results have been presented to illustrate the validity of our theoretical analysis and proposed algorithms.

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