

Impulse Response of sinc^N FIR Filters

S. C. Dutta Roy, *Fellow, IEEE*

Abstract—A simple recurrence formula is presented for computing the impulse response coefficients of the sinc^N FIR filter, consisting of a cascade of N sinc filters, each of length M . A closed form expression is also given for the first M coefficients.

Index Terms—Impulse response, sinc^N filters, digital filters.

I. INTRODUCTION

A sinc^N filter has the magnitude response

$$|H(e^{j\omega})| = \left| \frac{\sin(M\omega/2)}{M \sin(\omega/2)} \right|^N \quad (1)$$

and has the transfer function

$$H(z) = \left[\frac{(1 - z^{-M})}{M(1 - z^{-1})} \right]^N \quad (2)$$

$$= M^{-N} [1 + z^{-1} + z^{-2} + \dots + z^{-(M-1)}]^N. \quad (3)$$

As can be seen from (2) and (3), the sinc^N filter is a cascade of N sinc or moving average filters, each of length M . Such filters are useful in decimation and interpolation applications [1]–[11]. Several architectures are available for realizing these filters, of which the cascaded-integrator-comb (CIC) [1] and the nonrecursive FIR architectures [3], [4] are well known. The latter approach requires off-line calculation of all the impulse response coefficients. In a recent paper, Shiraishi [8] presented a generalization of the simultaneous coefficient calculation method for the second approach, which was earlier investigated for $N = 3$. For this purpose, he developed a multiple summation formula for the coefficients and also its recurrence form. In this paper, we present a simple recurrence formula for the coefficients, and also give a closed form expression for the first M coefficients.

II. RECURRENCE RELATION

For simplicity, we consider the scaled transfer function

$$G(z) = M^N H(z) = \left[\frac{(1 - z^{-M})}{(1 - z^{-1})} \right]^N. \quad (4)$$

Obviously, $G(z)$ is a polynomial of the form

$$G(z) = g_0 + g_1 z^{-1} + g_2 z^{-2} + \dots + g_{N(M-1)} z^{-N(M-1)}. \quad (5)$$

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The author is with the Department of Electrical Engineering, Indian Institute of Technology, New Delhi 110016, India (e-mail: scdroy@ee.iitd.ernet.in).

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Our problem is to find the g_n 's, $n = 0$ to $N(M-1)$. However, from (3), it is obvious that $G(z)$ is a symmetrical coefficient linear phase polynomial i.e.,

$$g_n = g_{N(M-1)-n}. \quad (6)$$

Thus, we need to find g_n 's for $n = 0$ to

$$n_0 = [N(M-1) - q]/2 \quad (7)$$

where $q = 1$ or 0 depending on whether $N(M-1)$ is odd or even.

Now from (4), (5) and the binomial theorem, we get

$$1 - \binom{N}{1} z^{-M} + \binom{N}{2} z^{-2M} - \binom{N}{3} z^{-3M} \\ + \dots + (-1)^N \binom{N}{N} z^{-NM} \\ = \left[1 - \binom{N}{1} z^{-1} + \binom{N}{2} z^{-2} - \binom{N}{3} z^{-3} \right. \\ \left. + \dots + (-1)^N \binom{N}{N} z^{-N} \right] \\ \times [g_0 + g_1 z^{-1} + g_2 z^{-2} \\ + \dots + g_{N(M-1)} z^{-N(M-1)}]. \quad (8)$$

By performing the multiplication on the right-hand side of (8), we get (9) as shown at the bottom of the next page, where, it can be observed, the same order terms on the right-hand side are located on a line slanting to the left. In equating the coefficients of z^{-n} on both sides, we can easily see that there are two different cases. For $n = rM$, where r is an integer and $1 \leq r \leq N$, the sum of the coefficients of z^{-n} on the right-hand side must be $(-1)^r \binom{N}{r}$. Thus

$$\sum_{i=0}^p (-1)^i \binom{N}{i} g_{n-i} = (-1)^r \binom{N}{r} \quad (10)$$

where

$$p = \min(N, n) \quad (11)$$

i.e., the last term in the left-hand side summation in (10) contains either $\binom{N}{N}$ or g_0 . On the other hand, for $n \neq rM$, the sum of the coefficients of z^{-n} on the right-hand side must be zero, i.e.,

$$\sum_{i=0}^p (-1)^i \binom{N}{i} g_{n-i} = 0. \quad (12)$$

TABLE I
VALUES OF $g(n)$ FOR $N = 2, 3, 4$ AND $M = 2$ TO 8

N=2															
M	g_0	g_1	g_2	g_3	g_4	g_5	g_6	g_7	g_8	g_9	g_{10}	g_{11}	g_{12}	g_{13}	g_{14}
2	1	2													
3	1	2	3												
4	1	2	3	4											
5	1	2	3	4	5										
6	1	2	3	4	5	6									
7	1	2	3	4	5	6	7								
8	1	2	3	4	5	6	7	8							
N=3															
2	1	3													
3	1	3	6	7											
4	1	3	6	10	12										
5	1	3	6	10	15	18	19								
6	1	3	6	10	15	21	25	27							
7	1	3	6	10	15	21	28	33	37						
8	1	3	6	10	15	21	28	36	46	48					
N=4															
2	1	4													
3	1	4	10	16	19										
4	1	4	10	20	31	40	44								
5	1	4	10	20	35	52	68	80	85						
6	1	4	10	20	35	56	80	104	125	140	146				
7	1	4	10	20	35	56	84	116	149	180	206	224	231		
8	1	4	10	20	35	56	84	120	161	204	246	284	315	336	344

also given in [8], the computational effort is considerably reduced. In fact, the computational complexity then becomes somewhat comparable to that of the method given in this paper. However, a quantitative comparison of the computational effort is difficult to formulate because of the difference in the nature of the computations. Our method has the advantages of simplicity, compactness and explicitness, not only for the first M impulse response coefficients, but in the computation of the rest of the coefficients also. Both methods can be easily implemented in computer software or hardware. Also, in either method, there are no potential numerical problems.

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