

CALCULATING THE BIT-SAVING ACHIEVED WITH NOISE-SHAPING

The usual (first-order) “noise shaping” filter [1,2] is of the form:

$$H(z) = 1 - z^{-1} \quad (1)$$

which is a simple high-pass with a zero at $z=1$. In the case of an k^{th} order noise shaper, we have $H(z)^k$. For practical purposes, only $k=1$ and $k=2$ (as achieved with the noise-shaping structure) are likely to be sufficiently stable to be useful.

The magnitude of the frequency response of $H(z)$ is just:

$$|H(z)| = [(1 - e^{-j\omega})(1 - e^{j\omega})]^{1/2} = [2 - 2\cos(\omega)]^{1/2} \quad (2)$$

Because we will be interested in how a noise shaping filter will modify the power in the spectrum, we find it convenient to work with the squared magnitude of $H(z)$:

$$|H(z)|^2 = [(1 - e^{-j\omega})(1 - e^{j\omega})] = [2 - 2\cos(\omega)] = 4 \sin^2(\omega/2) \quad (3)$$

which for a k^{th} order noise shaper is:

$$|H(z)|^{2k} = [(1 - e^{-j\omega})(1 - e^{j\omega})]^k = [2 - 2\cos(\omega)]^k = 4^k \sin^{2k}(\omega/2) \quad (4)$$

Fig. 1 shows the relevant functions. Now, we are interested in finding the area under this curve from $\omega=0$ to some frequency that is π divided by the oversampling factor. If we have m octaves of oversampling, this upper frequency is $\pi/2^m$.

Here we will find it convenient to approximate the sine by its argument because we are interested mainly in fairly large oversampling factors. Thus we want the integral:

$$p = \int_0^{\pi/2^m} 4^k \sin^{2k}(\omega/2) d\omega \approx \int_0^{\pi/2^m} 4^k (\omega/2)^{2k} d\omega = (\pi/2^m)^{(2k+1)} / (2k+1) \quad (5)$$

This we want to compare to the non-oversampling, no-noise-shaping case.

$$p_0 = \int_0^{\pi} 1 d\omega = \pi \quad (6)$$

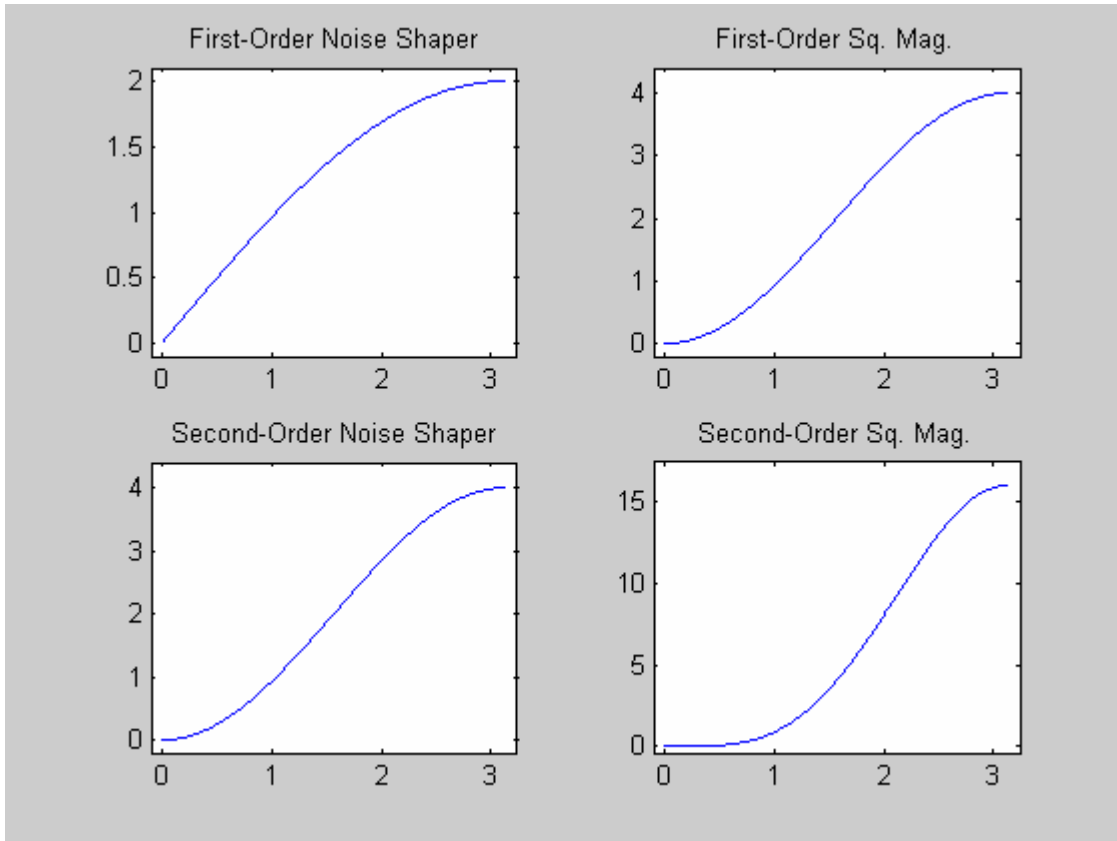


Fig. 1 First- and Second-Order Noise Shaping Curves

so we have p/p_0 as:

$$p/p_0 = \pi^{2k} / [(2^{m(2k+1)})(2k+1)] \quad (7)$$

Now, this reduction of noise power could also have been achieved if we had added ΔB bits of resolution to the signal, which would reduce the noise power by $2^{-2\Delta B}$, so

$$2^{-2\Delta B} = p/p_0 = \pi^{2k} / [(2^{m(2k+1)})(2k+1)] \quad (8)$$

Taking the log base 2 of both sides:

$$-2\Delta B = \text{Log}_2[\pi^{2k}/(2k+1)] - \text{Log}_2[2^{m(2k+1)}] \quad (9)$$

or:

$$\begin{aligned} \Delta B &= (1/2) \text{Log}_2[2^{m(2k+1)}] - (1/2) \text{Log}_2[\pi^{2k}/(2k+1)] \\ &= m(k+1/2) - (1/2) \text{Log}_2[\pi^{2k}/(2k+1)] \end{aligned} \quad (10)$$

This is the details of a result quoted in Orfanidis [1] and given in part in a previous Application Note No. 345 [2].

{Note: Please check any copies of AN-345 you may have. We have come across copies that have $\text{Log}_2 m$ instead of m in equation (9) of that AN, equivalent to equation (10) here. We are defining m as “octaves.” If it were the oversampling factor here, then the Log_2 would be correct. While our originals and the copies we are currently shipping are correct, it is possible that a few copies were made from an earlier incorrect version, for which we apologize.}

We mentioned that Orfanidis used a reasonable approximation of $\sin(x) \approx x$ for small x . Because we really are interested in only $k=1$ and $k=2$, we can do the exact problem using tabulated integrals [3].

$$p = \int_0^{\pi/2^m} 4^k \sin^{2k}(\omega/2) d\omega \quad (11)$$

For which the $k=1$ case gives us:

$$p_1 = \int_0^{\pi/2^m} 4 \sin^2(\omega/2) d\omega = 2 [\pi/2^m - \sin(\pi/2^m)] \quad (12)$$

while the $k=2$ case gives us:

$$p_2 = \int_0^{\pi/2^m} 4^2 \sin^4(\omega/2) d\omega = 6\pi/2^m - 8\sin(\pi/2^m) + \sin(2\pi/2^m) \quad (13)$$

Accordingly we have only to plug in the value of m to $p = p_1$ or p_2 , and calculate the added bits as :

$$\Delta B = -(1/2) \log_2(p/\pi) \quad (14)$$

Table 1 shows the results of calculations using the exact formulas, equations (12) and (13), the Orfanidis formula, equation (10) and the “Hauser Rule of Thumb” [4]. [This rule is that first-order gives 1.5 bits/octave with a one-bit penalty, while second order gives us 2.5 bits/octave with a 2 bit penalty.] Since most practical cases will involve at least a factor of 16 of oversampling, we see excellent agreement regardless of the formulas used.

TABLE 1

Bit Savings for First- and Second-Order Noise Shaping
for m Octaves of Oversampling

Octaves of Oversampling ↓ ↓ ↓	<u>First-Order Noise Shaping</u>			<u>Second-Order Noise Shaping</u>		
	Exact	Orfanidis	Hauser	Exact	Orfanidis	Hauser
	m=0	-0.5000	-0.8590	-1.0000	-2.1420	0
m=1	0.7302	0.6410	0.5000	0.5704	0.5704	0.5000
m=2	2.1632	2.1410	2.0000	2.9110	2.9110	3.0000
m=3	3.6465	3.6410	3.5000	5.3712	5.3712	5.5000
m=4	5.1424	5.1410	5.0000	7.8613	7.8613	8.0000
m=5	6.6413	6.6410	6.5000	10.3588	10.3588	10.5000
m=6	8.1411	8.1410	8.0000	12.8582	12.8582	13.0000
m=7	9.6410	9.6410	9.5000	15.3580	15.3580	15.5000
m=8	11.1410	11.1410	11.0000	17.8580	17.8580	18.0000
m=9	12.6410	12.6410	12.5000	20.3580	20.3580	20.5000
m=10	14.1410	14.1410	14.0000	22.8580	22.8580	23.0000

REFERENCES

- [1] B. Hutchins, "Improved Signal/Noise Ratio with First-Order Noise Shaping: An Example," Electronotes Application Note No. 345, October 1997
- [2] S. Orfanidis, Introduction to Signal Processing, Prentice-Hall (1996) pp 67-73
- [3] CRC Standard Mathematical Tables, 25th Ed., CRC Press (1978)
- [4] M. Hauser, "Principles of Oversampling A/D Conversion," J. Audio Eng. Soc., Vol. 39, No. 1/2, Jan/Feb 1991, pp 3-26.