

F

Goertzel Algorithm

Goertzel's algorithm performs a DFT using an IIR filter calculation. Compared to a direct N -point DFT calculation, this algorithm uses half the number of real multiplications, the same number of real additions, and requires approximately $1/N$ the number of trigonometric evaluations. The biggest advantage of the Goertzel algorithm over the direct DFT is the reduction of the trigonometric evaluations. Both the direct method and the Goertzel method are more efficient than the FFT when a "small" number of spectrum points is required rather than the entire spectrum. However, for the entire spectrum, the Goertzel algorithm is an N^2 effort, just as is the direct DFT.

F.1 DESIGN CONSIDERATIONS

Both the first order and the second order Goertzel algorithms are explained in several books [1–3] and in Ref. 4. A discussion of them follows. Since

$$W_N^{-kN} = e^{j2\pi k} = 1$$

both sides of the DFT in (6.1) can be multiplied by it, giving

$$X(k) = W_N^{-kN} \sum_{r=0}^{N-1} x(k) W_N^{r+kr} \quad (\text{F.1})$$

which can be written

$$X(k) = \sum_{r=0}^{N-1} x(r) W_N^{-k(N-r)} \quad (\text{F.2})$$

Define a discrete-time function as

$$y_k(n) = \sum_{r=0}^{N-1} x(r) W_N^{-k(n-r)} \quad (\text{F.3})$$

The discrete transform is then

$$X(k) = y_k(n)|_{n=N} \quad (\text{F.4})$$

Equation (F.3) is a discrete convolution of a finite-duration input sequence $x(n)$, $0 < n < N - 1$, with the infinite sequence W_N^{-kn} . The infinite impulse response is therefore

$$h(n) = W_N^{-kn} \quad (\text{F.5})$$

The z -transform of $h(n)$ in (F.5) is

$$H(z) = \sum_{n=0}^{\infty} h(n) z^{-n} \quad (\text{F.6})$$

Substituting (F.5) into (F.6) gives

$$H(z) = \sum_{n=0}^{\infty} W_N^{-kn} z^{-n} = 1 + W_N^{-k} z^{-1} + W_N^{-2k} z^{-2} + \cdots = \frac{1}{1 - W_N^{-2k} z^{-1}} \quad (\text{F.7})$$

Thus, equation (F.7) represents the transfer function of the convolution sum in equation (F.3). Its flow graph represents the first order Goertzel algorithm and is shown in Figure F.1. The DFT of the k th frequency component is calculated by

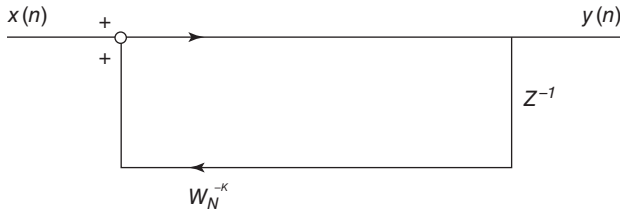


FIGURE F.1. First order Goertzel algorithm.

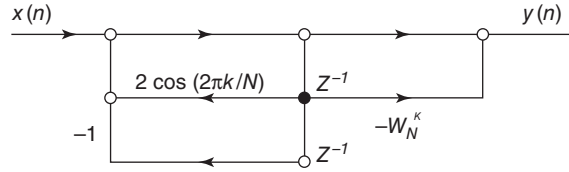


FIGURE F.2. Second order Goertzel algorithm.

starting with the initial condition $y_k(-1) = 0$ and running through N iterations to obtain the solution $X(k) = y_k(N)$. The $x(n)$'s are processed in time order, and processing can start as soon as the first one comes in. This structure needs the same number of real multiplications and additions as the direct DFT but $1/N$ the number of trigonometric evaluations.

The second order Goertzel algorithm can be obtained by multiplying the numerator and denominator of (F.7) by $1 - W_N^{-kn} z^{-1}$ to give

$$H(z) = \frac{1 - W_N^{+k} z^{-1}}{1 - 2 \cos(2\pi k/N) z^{-1} + z^{-2}} \quad (\text{F.8})$$

The flow graph for this equation is shown in Figure F.2. Note that the left half of the graph contains feedback flows and the right half contains only feedforward terms. Therefore, only the left half of the flow graph must be evaluated each iteration. The feedforward terms need only be calculated once for $y_k(N)$. For real data, there is only one real multiplication in this graph and only one trigonometric evaluation for each frequency. Scaling is a problem for fixed-point arithmetic realizations of this filter structure; therefore, simulation is extremely useful.

The second order Goertzel algorithm is more efficient than the first order Goertzel algorithm. The first order Goertzel algorithm (assuming a real input function) requires approximately $4N$ real multiplications, $3N$ real additions, and two trigonometric evaluations per frequency component as opposed to N real multiplications, $2N$ real additions, and two trigonometric evaluations per frequency component for the second order Goertzel algorithm. The direct DFT requires approximately $2N$ real multiplications, $2N$ real additions, and $2N$ trigonometric evaluations per frequency component.

This Goertzel algorithm is useful in situations where only a few points in the spectrum are necessary, as opposed to the entire spectrum. Detection of several discrete frequency components is a good example. Since the algorithm processes samples in time order, it allows the calculation to begin when the first sample arrives. In contrast, the FFT must have the entire frame in order to start the calculation.

REFERENCES

1. G. Goertzel, An algorithm for the evaluation of finite trigonometric series, *American Mathematics Monthly*, vol. 65, Jan. 1958.
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3. C. S. Burrus and T. W. Parks, *DFT/FFT and Convolution Algorithms: Theory and Implementation*, Wiley, Hoboken, NJ, 1988.
4. http://ptolemy.eecs.berkeley.edu/papers/96/dtmf_ict/www/node3.html.