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DESIGN AND MICROPROCESSOR-BASED IMPLEMENTATION
 OF DIGITAL FREQUENCY SELECTIVE FILTERS FOR
 APPLICATION IN
 TERRAIN ROUGHNESS IDENTIFICATION



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20. ABSTRACT (Continue on reverse side if necessary and identify by block number) The design and implementation of microprocessor-based frequency selective filters for possible use in an on-board terrain roughness identification scheme are investigated. In the design aspect of the investigation, a systematic digital filter design procedure is developed. The effectiveness of the procedure is illustrated by several digital filter design examples. In the implementation aspect, a digital signal processing firmware (hardware and software) based on the Motorola 6802 microprocessor is developed. The		

microprocessor realization of the exemplary digital filters are carried out. Actual experimental results are recorded and compared to their theoretical simulations. The capability of the firmware which includes the computational speed and numerical accuracies will be shown to be adequate for the present purpose.

PREFACE

Technical advances in the on-the-move adjustability of military vehicle suspension components, on board terrain sensing, modern system control theory, and microprocessors have combined in recent years to greatly increase the potential for improving the ride performance of military vehicles. Increasing emphasis on fire-on-the move, lighter weight combat vehicles, and higher horsepower per ton ratios make the role of the suspension system more critical for mission performance. This report documents and develops the theory and methods required for real time processing of sensed terrain elevation data in order to make it useful for suspension adjustment decisions. It also demonstrates the successful real time application of the techniques on a currently available microprocessor.

This work was performed for the Tank-Automotive Systems Laboratory of the U.S. Army Tank-Automotive Command, Warren, Michigan, under the overall direction of Mr. Michael Kaifesh, Chief of the Track and Suspension sub-function, and Mr. Leonard Sloncz, Track and Suspension project engineer. Mr. Robert Daigle of the Applied Research Function was technical monitor for the contract.

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1.0 INTRODUCTION

The possibility of obtaining improvement in the ride quality of a vehicle using a damping rate which varies according to the terrain roughness has been considered in literatures such as [1] - [5]. Recently, a preliminary feasibility study on the identification of terrain roughness and frequency characteristics was carried out by Daigle [5] using sampled-data analysis and digital filtering techniques. In this study, a mathematical decision scheme for characterizing the terrain roughness in terms of its frequency (wave lengths) contents is developed. The development assumes that the terrain elevation can be sensed, sampled and digitized . The terrain elevation sampled-data stream is then passed through frequency selective filters which separate the frequency components of the data into different adjacent bands or channels on the frequency spectrum. The RMS value from each channel is computed, and the relative amplitude of the RMS values is used to indicate certain degrees of terrain roughness present in each of the channels.

Simulations of the above sampled-data, digital filtering and decision scheme were made on the Systems Engineering Laboratories (SEL) digital computer. The effectiveness of the scheme for indicating or identifying the terrain roughness was strongly supported by the simulation results. The use of the terrain roughness identification scheme is proposed [5], among other techniques, as a possible means of incorporating a microprocessor-based on-board adaptive suspension control unit for a vehicle.

To determine the feasibility of an actual implementation of the microprocessor-based on-board system, a preliminary investigation into experimental microprocessor-based filters is suggested. A main concern of the investigation is the computational speed and numerical accuracies of the microprocessor in the realization of high order digital filters. As a rough guideline, it is noted that the digital frequency selective filters used in the formulation of the above terrain roughness

identification scheme consist of bandpass and highpass filters whose critical frequencies are less than 10 Hz.

2.0 OBJECTIVES

The objective of this report is to investigate the design and to carry out the actual implementation of microprocessor-based frequency selective filters which may be suitable for the terrain roughness identification purposes. In the design phase of the investigation, a systematic procedure for designing digital filters is developed. The procedure is based on bilinear transformation technique with emphasized consideration on the compensation of frequency warping and on the choice of the ratio of working frequency to sampling frequency. The effectiveness of the proposed design procedure will be demonstrated by several examples. It is remarked that the potential of the procedure may be enhanced by the incorporation of computer-aided digital filter design techniques.

In the implementation phase of the investigation, the hardware and software for a microprocessor-based digital signal processing system will be developed. Microprocessor realization of digital frequency selective filters will be demonstrated by the implementation of digital lowpass, highpass, bandpass and bandstop filters. The actual experimental frequency responses of the microprocessor-based digital filters will be recorded and compared to their theoretical frequency responses.

The organization of this report is as follows. The systematic procedure for the design of digital filters is developed in Section 5.1, the hardware and software for the microprocessor-based signal processing system and filters is described in Section 5.2 and Appendix B. The actual experimental frequency response of the microprocessor-based digital filters is given in Section 5.3. Section 4 discusses the results of the investigation and provides a few recommendations for the direction of future effort. A summary on the design of analog Butterworth frequency selective filters is given in Appendix A.

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3.0 CONCLUSIONS

The design procedure developed in Section 5 provides a systematic technique for obtaining a digital filter from a corresponding analog filter using bilinear transformation. The technique takes into consideration the frequency warping and the ratio of working to sampling frequencies. The effectiveness of the design procedures is illustrated by Examples 1 - 5, where the design specifications are satisfactorily fulfilled.

Based on the experimental results and performance of the microprocessor-based frequency selective filters presented in Section 5, the following may be inferred:

- The computational speed of the microprocessor-based system is sufficiently fast for the implementation of the digital frequency selective filters with the required specifications. As noted in the Introduction, the critical frequencies of the digital filters required in the terrain roughness identification schemes are less than 10 Hz. It is seen in Table 2 that the critical frequencies of the experimental microprocessor-based filters can be much higher than the required specification.* This further implies that there is room in the processing time for implementing higher order filters.
- The numerical accuracies of the microprocessor-based system using 12-bit word length data is adequate for the implementation of the filters. This is clearly illustrated by comparing the theoretical frequency responses of the digital filters depicted in Figs. 6, 8, 10, 12 and 14 to the actual experimental frequency responses of the microprocessor-based filters depicted by Figs. 19, 20, 21, 22 and 23.

* It is reminded that the microprocessor-based filters can readily be tuned, by adjusting the sampling frequency ω_s , so that the critical frequencies coincide with the desired specifications.

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4.0 RECOMMENDATIONS

The successful preliminary investigation into the microprocessor realization of the digital frequency selective filters provides a favorable possibility for implementing a microprocessor-based on-board terrain roughness identification system using digital filtering techniques. The following effort in line with the investigation of microprocessor-based signal processing system in this report may be pursued in the future:

- . Implementation of microprocessor-based system with parallel processing;
- . Use of 16-bit microprocessors;
- . Use of fast arithmetic and support chips;
- . Computer aided design package for the design procedures developed in Section 5
- . Implementation of the mathematical decision criterion for identifying the terrain roughness and frequency content as suggested in [5].

Some of these efforts are currently underway.

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5.0 DISCUSSION

5.1 DESIGN OF DIGITAL FILTERS

The basic steps in the design of digital filters generally involve:

- (i) the specification on the general characteristics of the filters;
- (ii) the approximation and design consideration in attaining the specification;
- (iii) the realization of the filters using finite precision arithmetic.

Step (i) depends mainly on the application of the filters while step (ii) depends on the design approach adopted by the designer. Step (iii) takes into account the limitations of digital devices, such as the finite word length in a digital circuit or machine and the finite computational speed.

The objective of this section is concerned with step (ii) of the design; step (iii) will be considered in the next section. In particular, the design of infinite impulse response (IIR) digital filters (lowpass, highpass, bandpass and bandstop filters) using analog filter formulas and bilinear transformation will be presented in detail in this section.

While a digital filter may directly be designed using pole-zero placement technique in the z -plane, a more traditional approach is to transform an analog filter, based on the poles and zeroes in the s -plane, into a corresponding digital filter satisfying a prescribed specification. Some of the reasons for the latter approach include :

- . the straightforward convergence of frequency specification (in terms of Hz or rad-s^{-1}) for an analog filter into the frequency specification (in terms of radian frequency, angle around the unit circle, or ratio of frequencies) for the digital filter, once the sampling rate is given;
- . the utilization of the highly developed art in the design of a variety of analog filters to obtain the corresponding digital filters (e.g. Butterworth, Chebychev or elliptic filters)
- . the closed-form design formulas for analog filters which can be translated

to yield closed-form design formulas for the corresponding digital filters. The closed-form formulas facilitate simplicity in the realization of the filters.

There are many techniques for transforming or converting an analog filter into a corresponding digital filter. One such technique is the bilinear transformation which is described below.

5.1.1 DESIGNING DIGITAL FILTERS FROM ANALOG FILTERS USING BILINEAR TRANSFORMATION WITH WORKING TO SAMPLING FREQUENCY RATIO CONSIDERATION

The design of digital filters from application of bilinear transformation to the formulas of analog filters has been considered in literature such as [6]-[7]. Most of the design procedures in these literature, however, do not include a systematic way for determining the gain (say, τ) in the bilinear transformation. As will be seen shortly, the transformation gain τ is closely related to the quality in the zero-order-hold reproduction of a processed sampled data from an analog signal by a digital filter. To ensure a desirable reproduction quality in the digital filter output, it is important that a proper transformation gain is used in the design.

In this section, we present a systematic approach to the design of digital filters from analog filters using the bilinear transformation which takes the ratio of working or critical frequencies of the digital filter to the sampling frequency into the design consideration. The approach provides a straightforward procedure for choosing the transformation gain τ and for obtaining the desirable output reproduction quality in the digital filter. The procedure is also well suited for use in computer-aided digital filter designs.

BILINEAR TRANSFORMATION

For sampled-data signals, the Laplace transform (s-transform) can be shown to be related to the z-transform by

$$z = e^{sT} , \quad (1)$$

where T is the sampling period. Using Pade's approximation [8], (1) can be approximated by

$$e^{sT} \approx \frac{1 + sT/2}{1 - sT/2} . \quad (2)$$

In general, one may redefine the mapping as

$$z \triangleq \frac{1 + s\tau/2}{1 - s\tau/2} \quad (3a)$$

or

$$s = \frac{2}{\tau} \frac{z - 1}{z + 1}, \quad (3b)$$

where τ is the transformation gain. Relationship (3) is known as BILINEAR TRANSFORMATION.

The mapping of the s-plane into the z-plane by bilinear transformation (s) is shown in Fig. 1, which can be constructed using the following relationships.

Define (see also Fig. 1)

$$\begin{aligned} \omega_a &\triangleq \text{analog frequency (rad-s}^{-1}\text{)}, \\ \omega_d &\triangleq \text{corresponding digital frequency (rad-s}^{-1}\text{)}, \\ \omega_s &\triangleq 2\pi/T = \text{sampling frequency (rad-s}^{-1}\text{)}, \\ \theta &\triangleq \frac{\omega_d}{\omega_s} 2\pi = \omega_d T = \text{radian frequency (rad)}. \end{aligned}$$

(a) Using (3a), the frequency axis of the s-plane (the imaginary axis, $s = j\omega_a$) is mapped into that of the z-plane (the unit circle, $z = e^{j\theta}$) as follows:

$$\begin{aligned} z &= \left. \frac{1 + s\tau/2}{1 - s\tau/2} \right|_{s=j\omega_a} \\ &= \frac{1 + j\omega_a\tau/2}{1 - j\omega_a\tau/2} \\ &= \frac{\sqrt{[1 + (\omega_a\tau/2)^2]} e^{j\tan^{-1}(\omega_a\tau/2)}}{\sqrt{[1 + (\omega_a\tau/2)^2]} e^{j\tan^{-1}(-\omega_a\tau/2)}} \end{aligned}$$

$$= e^{j2\tan^{-1}(\omega_a \tau/2)}$$

$$\stackrel{\Delta}{=} e^{j\theta} \quad (4)$$

Since $\theta = \frac{\omega_d}{\omega_s} 2\pi = \omega_d T$, (4) yields

$$\omega_d T = 2\tan^{-1}(\omega_a \tau/2) \quad (5a)$$

or

$$\omega_a = \frac{2}{\tau} \tan\left(\frac{\omega_d}{\omega_s} \pi\right) . \quad (5b)$$

The relationship (5) represents the frequency warping or distortion of bilinear transformation. The characteristic of the distortion is depicted in Figs. 2 and 3.

(b) From (3a), the real axis ($s = \sigma$) of the s-plane is mapped into the z-plane as the magnitude of

$$z = \frac{1 + \sigma\tau/2}{1 - \sigma\tau/2} , \quad (6)$$

where σ is real, and where it is seen that

$$-1 < z < 1 \text{ for } -\infty < \sigma < 0$$

$$1 \leq z < \infty \text{ for } 0 \leq \sigma < 2/\tau$$

$$-\infty < z < -1 \text{ for } \frac{2}{\tau} < \sigma < \infty .$$

It is clear from the above that the left half of the s-plane is mapped into the unit disk of the z-plane.

Now, let $G(s)$ denote the transfer function of an analog filter and $G(z)$ denote that of a corresponding digital filter. Then using bilinear transformation (3), the digital filter can be obtained as

$$G(z) = G(s) \left| \begin{array}{l} s = \frac{2}{\tau} \frac{z - 1}{z + 1} \end{array} \right. \quad (7)$$

By the mapping of the bilinear transformation (Fig. 1), all the stable poles of $G(s)$ will be converted into stable poles in $G(z)$. Consequently, the bilinear transformation (7) always yield stable digital filters from stable analog filters.

The effect of frequency warping or distortion (5) on the analog to digital filter conversion (7) is illustrated in Fig. 3. The figure also clearly reveals an explanation for the phenomenon of aliasings.

It is remarked that the transformation gain τ for the bilinear transformation (7) has not been specified. A systematic technique for determining τ is given in the sequel. The technique also automatically compensates for the frequency warping or distortion.

DESIGN CONSIDERATION

The digital zero-order-hold (ZOH) reproduction of an analog signal having a dominant working frequency ω_a (or correspondingly ω_d) depends on the ratio ω_d/ω_s . Fig. 4 illustrates the variation, with respect to the ratio ω_d/ω_s , by a sample and ZOH scheme¹. As can be seen from the figure, the "quality" of the digital reproduction of the analog signal improves with lower ratio of ω_d/ω_s . It is, therefore, desirable to design a digital filter whose working or critical frequencies are much lower than the sampling frequencies. A first design consideration in a digital filter design is to ensure that the ratio

$$0 < \omega_d/\omega_s \ll .5 \quad (8)$$

¹ The microprocessor-based sample and ZOH scheme is described in Section 3.

Remark 1: For simplicity and clarity, the above argument is approached from time-domain point of view using visual experimental results. It may be remarked that similar conclusions can be obtained using frequency domain analysis [6]. One also notes that (8) is in agreement to the Sampling Theorem due to Shannon and Nyquist [6].

It is also important to observe the time delays in the digital outputs in reference to the continuous signals in Fig. 4.

Δ

Once the ratio ω_d/ω_s has been selected, one may define a factor R as

$$R \triangleq \tan \left(\frac{\omega_d}{\omega_s} \pi \right), \quad 0 < \omega_d/\omega_s < .5$$

$$= \frac{\omega_a \tau}{2}, \quad (9)$$

where (9) follows from (5). Note that $0 < R < \infty$. From (9), the transformation gain τ is obtained as

$$\tau = \frac{2R}{\omega_a}. \quad (10)$$

Using (8) in (3b), the bilinear transformation becomes

$$s = \frac{\omega_a}{R} \frac{z - 1}{z + 1}. \quad (11)$$

The above design consideration of first specifying a desired ratio ω_d/ω_s thus leads to a systematic choice of the transformation gain τ for the bilinear transformation as shown in (11).

Using (11) as the basis for bilinear transformation, the conversion of an analog filter $G(s)$ with working or critical frequency ω_a to a

corresponding digital filter $G(z)$ with working or critical frequency ω_d , follows from (7) as

$$G(z) = G(s) \left| \begin{array}{l} s = \frac{\omega_a}{R} \frac{z-1}{z+1} \end{array} \right. \quad (12)$$

The bilinear transformation in (12) ensures that the desired ω_d/ω_s will be obtained.

Finally, it is important to note that the working or critical frequency ω_d of the digital filter can be varied by simply adjusting the sampling frequency ω_s .

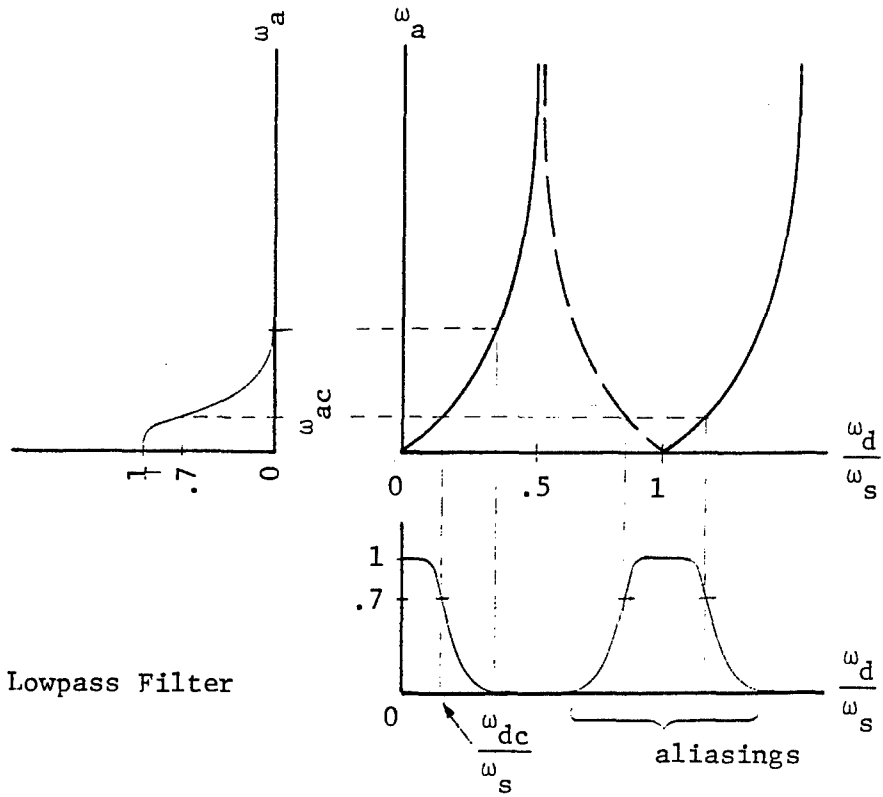


Fig. 3a) Lowpass Filter

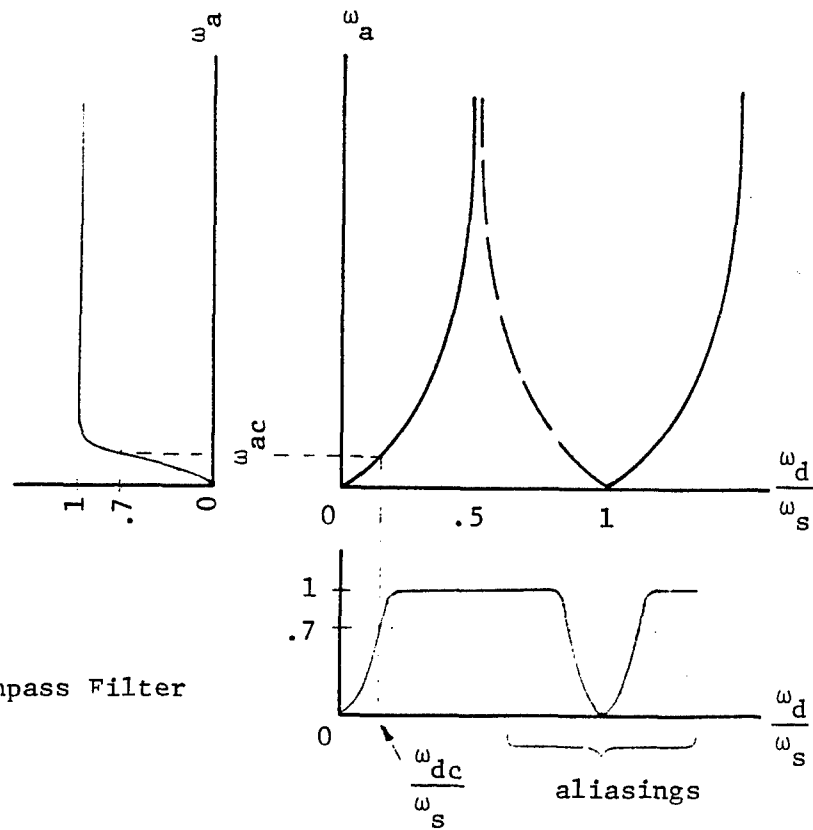


Fig. 3b) Highpass Filter

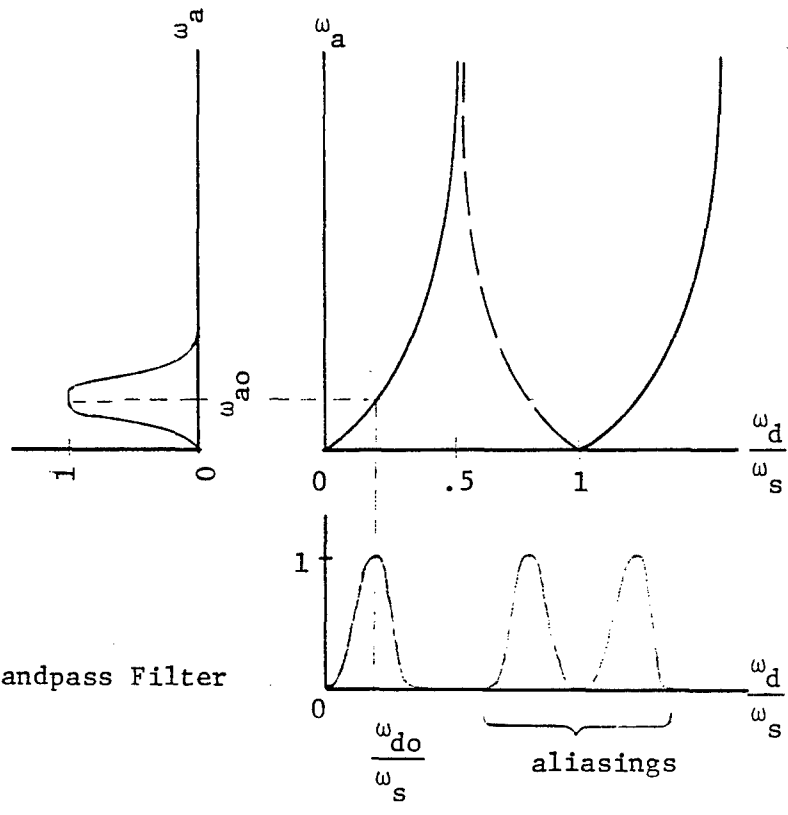


Fig. 3c) Bandpass Filter

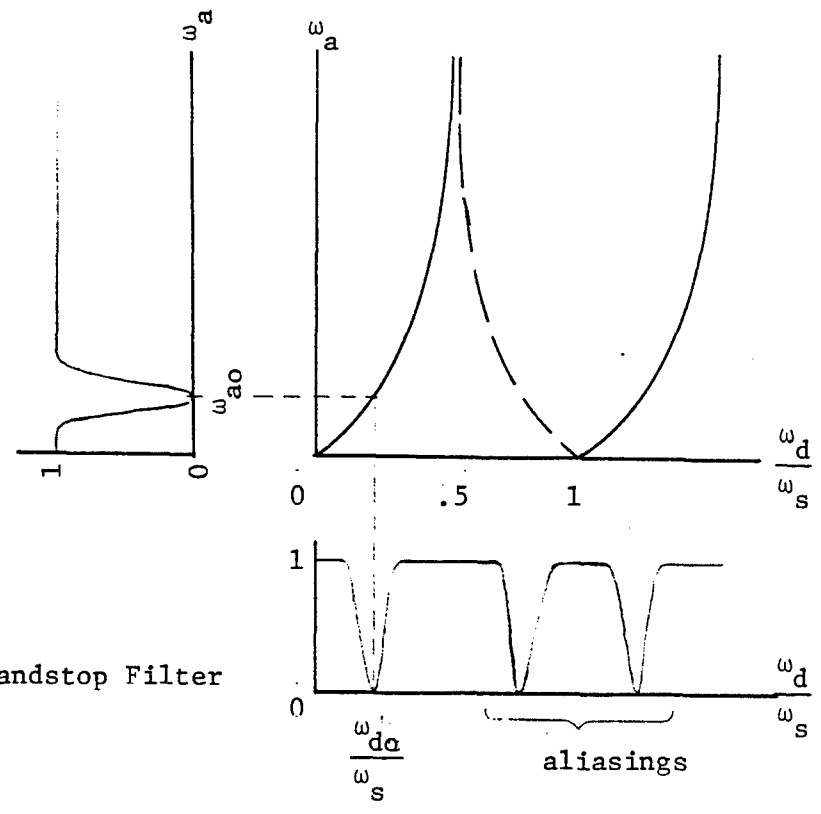
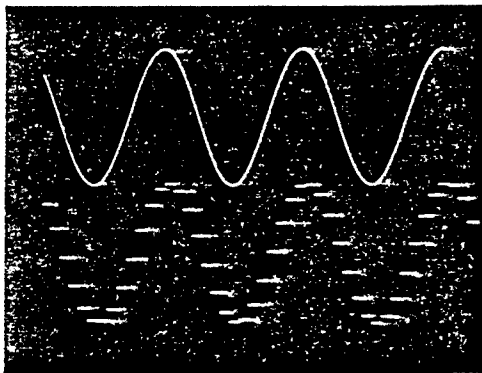
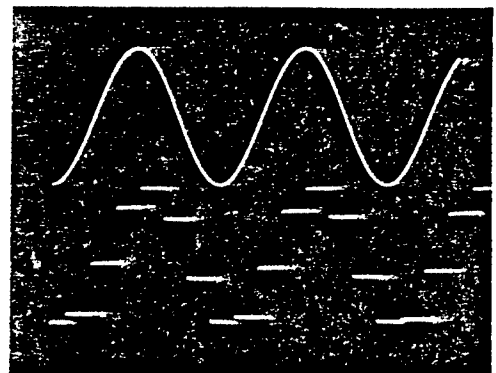


Fig. 3d) Bandstop Filter

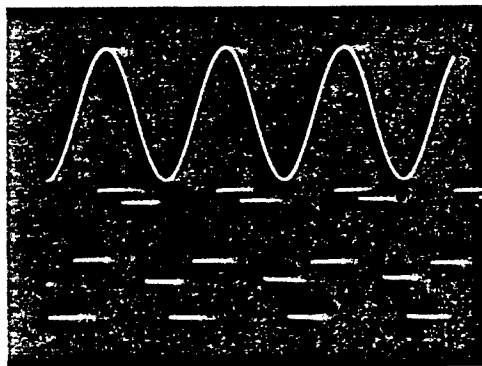
Fig. 3 Effects of Frequency Warping on Frequency Selective Filters



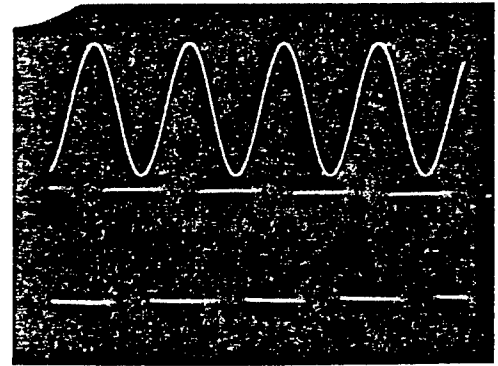
$$\omega_d/\omega_s = 1/15$$



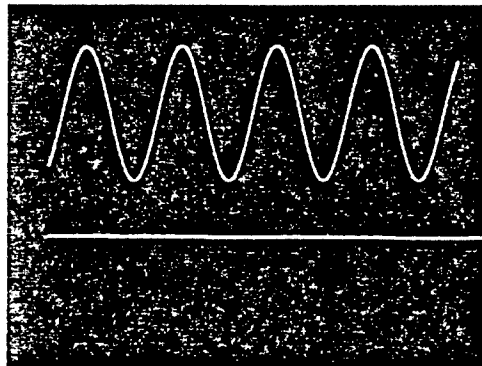
$$\omega_d/\omega_s = 1/7$$



$$\omega_d/\omega_s = 1/5$$



$$\omega_d/\omega_s = 1/2$$



$$\omega_d/\omega_s = 1$$

Fig.4. ZOH Reproduction of Sinusoidal Signal with respect to Ratio ω_d/ω_s .

5.1.2 DESIGN OF DIGITAL LOWPASS FILTERS

The design of an n-th order analog Butterworth lowpass filter $G_{LP}(s)$ with critical cut-off frequency ω_{ac} is given in Appendix A2. To obtain a corresponding digital Butterworth lowpass filter $G_{LP}(z)$, the following systematic procedure may be used:

- (a) Select the ratio ω_{dc}/ω_s , where ω_{dc} is the desired critical cut-off frequency of the digital lowpass filter (see Fig.3a).
- (b) Obtain the factor R and transformation gain as

$$\begin{aligned} R &= \tan \left(\frac{\omega_{dc}}{\omega_s} \pi \right) \\ &= \frac{\omega_{ac} \tau}{2} \end{aligned} \quad (13a)$$

or

$$\tau = \frac{2R}{\omega_{ac}} \quad (13b)$$

- (c) Using the substitution described by (12), a corresponding n-th order digital Butterworth lowpass filter is obtained as¹

$$\begin{aligned} G_{LP}(z) &= G_{LP}(s) \Big|_{s = \frac{\omega_{ac}}{R} \frac{z-1}{z+1}} \\ &= K \frac{(z+1)^n}{(z-p_1)(z-p_2) \dots (z-p_n)} \\ &\triangleq KG(z) \quad , \end{aligned} \quad (14a)$$

with

$$p_i \triangleq \frac{(1+u_i R)}{(1-u_i R)} \quad (14b)$$

and, for unity gain in the low passband,

¹ Further details in the manipulation are found in Appendix C.

$$K = \frac{R^n}{(1-u_1 R)(1-u_2 R) \dots (1-u_n R)}$$

$$= \frac{1}{G(1)}, \quad (14c)$$

where u_i are the poles of the normalized n-th order analog Butterworth lowpass filter $G_{LPN}(s)$ described in Appendix A1.

Remark 2:

- . Note that the digital filter is explicitly dependent on the factor R.
- . The critical frequency will be determined by the sampling frequency through the ratio ω_{dc}/ω_s .
- . The gain K may be arbitrarily chosen if so desired.

Δ

Example 1: 3rd Order Digital Butterworth Lowpass Filter

Problem: Design a 3rd order digital Butterworth lowpass filter with cut-off frequency ω_{dc} .

Solution: From Appendix A1, the normalized 3rd order analog Butterworth lowpass filter is given by

$$G_{LPN}(s) = \frac{1}{(s-u_1)(s-u_2)(s-u_3)}, \quad (15a)$$

where

$$u_1 = -1 \quad (15b)$$

$$u_2 = -.5 + j.866 \quad (15c)$$

$$u_3 = -.5 - j.866 \quad (15d)$$

Following the above procedures:

- (a) Select $\omega_{dc}/\omega_s = .1476$ or $\omega_s = 6.77\omega_{dc}$. With this ratio, the digital ZOH reproduction of a sine wave at ω_{dc} is approximately as shown in Fig.5 .

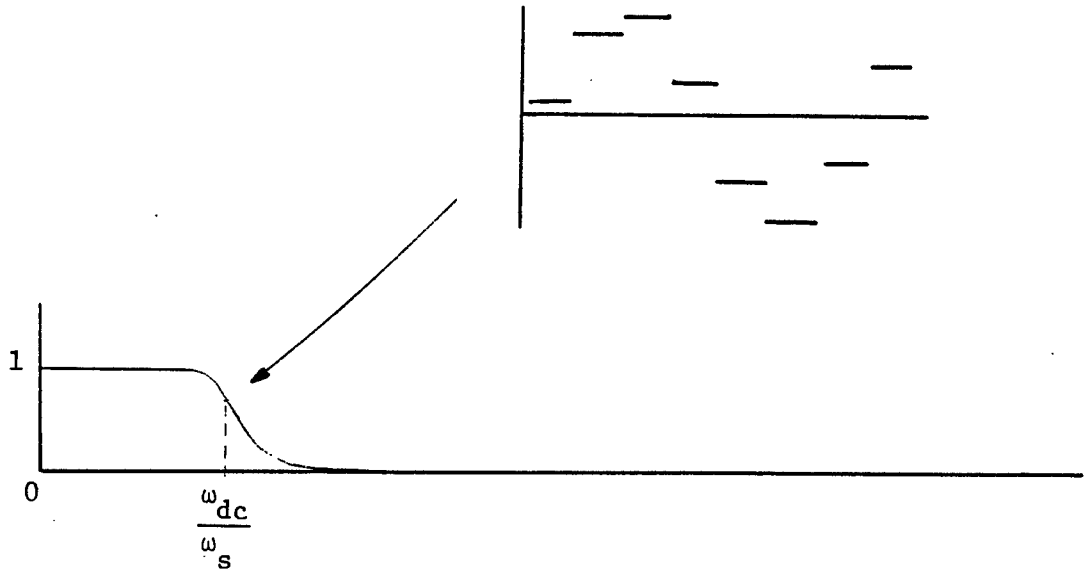


Fig.5

$$(b) \quad R = \tan(.1476\pi) = .5 \quad (\text{by choice of (a)}) \quad (16)$$

$$(c) \quad p_1 = \frac{1 + (-1)(.5)}{1 - (-1)(.5)} = .3333 \quad (17a)$$

$$p_2 = \frac{1 + (-.5 + j.866)(.5)}{1 - (-.5 + j.866)(.5)} = .4286 + j.4949 \quad (17b)$$

$$p_3 = \frac{1 + (-.5 - j.866)(.5)}{1 - (-.5 - j.866)(.5)} = .4286 - j.4949 = p_2^* \quad (17c)$$

$$K = \frac{(.5)^3}{(1 - (-1)(.5))(1 - (-.5 + j.866)(.5))(1 - (-.5 - j.866)(.5))}$$

$$= .04762 \quad (17d)$$

The 3rd order digital Butterworth lowpass filter is thus given by

$$G_{LP}(z) = \frac{.04762(z + 1)^3}{(z - .3333)(z - .4286 - j.4949)(z - .4286 + j.4949)}$$

$$= \frac{.04762(z^3 + 3z^2 + 3z + 1)}{z^3 - 1.1905z^2 + .7143z - .1429} \quad (18)$$

The theoretical frequency response of $G_{LP}(z)$ in (18), computed as $|G_{LP}(e^{j\omega T})|$ versus ω/ω_s , is shown in Fig. 6.

The recursive equation for the digital lowpass filter follows from (18) as

$$y(k) = 1.1905y(k-1) - .7143y(k-2) + .1429y(k-3)$$

$$+ .04762[u(k) + 3u(k-1) + 3u(k-2) + u(k-3)] \quad (19)$$

where $y(k)$ and $u(k)$ are respectively the discrete output and input sequences of the filter. The microprocessor-based implementation of the lowpass filter given by (18) or (19) is described in Section 3. An experimental frequency response of the microprocessor-based 3rd order digital Butterworth lowpass filter is presented in Section 4.

$$|G_{LP}(e^{j\omega_d T})|$$

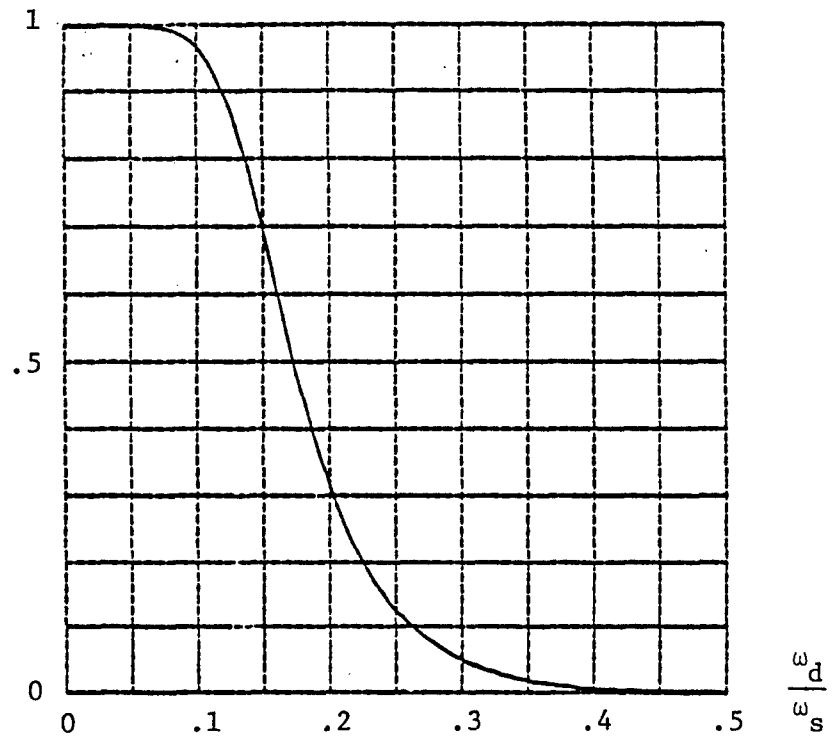


Fig. 6. Theoretical Frequency Response of $G_{LP}(z)$ in (18)

5.1.3 DESIGN OF DIGITAL HIGHPASS FILTERS

The design of an n-th order analog Butterworth highpass filter $G_{HP}(s)$ with critical cut-in frequency ω_{ac} is given in Appendix A3. The following procedure may be used to systematically obtain a corresponding digital Butterworth highpass filter $G_{HP}(z)$ from $G_{HP}(s)$.

- (a) Select the desired ratio ω_{dc}/ω_s where ω_{dc} is the critical cut-in frequency of the digital highpass filter (see Fig.3b).
 (b) Set

$$\begin{aligned} R &= \tan \left(\frac{\omega_{dc}}{\omega_s} \pi \right) \\ &= \frac{\omega_{ac} \tau}{2} \end{aligned} \quad (20a)$$

so that

$$\tau = \frac{2R}{\omega_{ac}} \quad (20b)$$

- (c) Using the conversion scheme (12), a corresponding n-th order digital Butterworth highpass filter is obtained as¹

$$\begin{aligned} G_{HP}(z) &= G_{HP}(s) \Big|_{s = \frac{\omega_{ac}}{R} \frac{z-1}{z+1}} \\ &= \frac{K(z-1)^n}{(z-p_1)(z-p_2) \dots (z-p_n)} \\ &\triangleq KG(z) \end{aligned} \quad (21a)$$

with

¹ Details given in Appendix C.

$$P_i = \frac{1 + u_i R}{1 - u_i R} \quad (21b)$$

and, for unity gain in the high passband,

$$K = \frac{1}{(1 - u_1 R)(1 - u_2 R) \dots (1 - u_n R)}$$

$$= \frac{1}{G(-1)} \quad (21c)$$

where u_i are the poles of the normalized n-th order analog Butterworth lowpass filter $G_{LPN}(s)$ described in Appendix A1.

Remark 2 similarly applies to the above design of digital Butterworth highpass filter.

Example 2: 3rd Order Digital Butterworth Highpass Filter

Problem: Design a 3rd order digital Butterworth highpass filter with cut-in frequency ω_{dc} .

Solution: From Appendix A3, the 3rd order normalized analog Butterworth lowpass filter is given by

$$G_{HPN}(s) = \frac{s^3}{(s - u_1)(s - u_2)(s - u_3)} \quad (22a)$$

where

$$u_1 = -1 \quad (22b)$$

$$u_2 = -.5 + j.866 \quad (22c)$$

$$u_3 = -.5 - j.866 \quad (22d)$$

Following the procedures outlined above:

- (a) Select $\omega_{dc}/\omega_s = .05$ or $\omega_s = 20 \omega_{dc}$. With this ratio, the digital ZOH reproduction of a sine wave at ω_{dc} is approximately as shown in Fig.7.

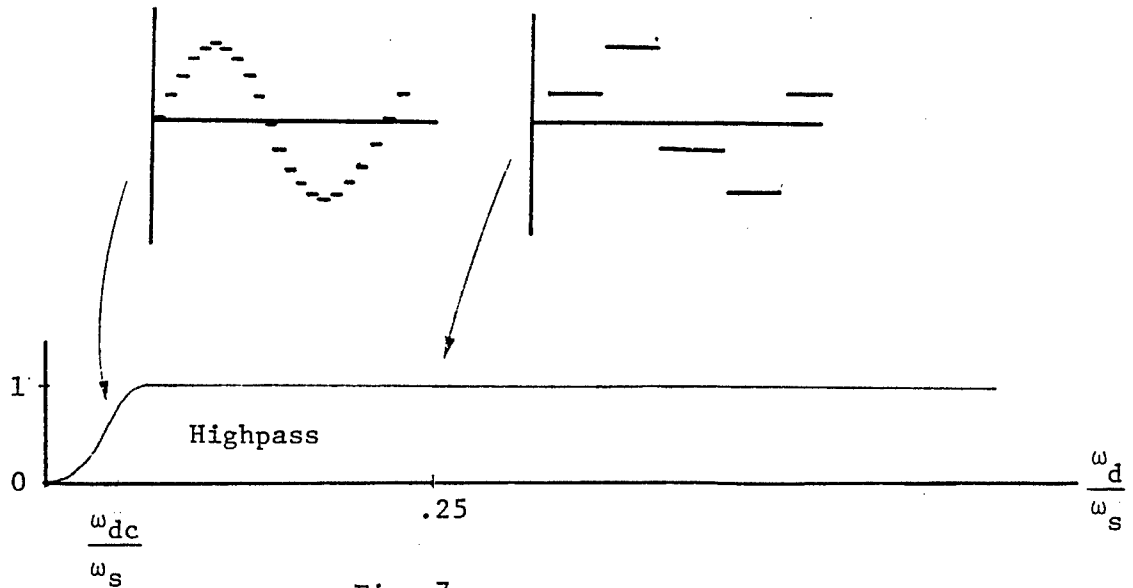


Fig. 7.

$$(b) R = \tan(.05\pi) = .1585 \quad (23)$$

$$(c) p_1 = \frac{1 + (-1)(.1585)}{1 - (-1)(.1585)} = .72637 \quad (24a)$$

$$p_2 = \frac{1 + (-.5 + j.866)(.1585)}{1 - (-.5 + j.866)(.1585)} = .82366 + j.23195 \quad (24b)$$

$$p_3 = \frac{1 + (-.5 - j.866)(.1585)}{1 - (-.5 - j.866)(.1585)} = .82366 - j.23195 = p_2^* \quad (24c)$$

Hence,

$$G_{HP}(z) = \frac{K(z-1)^3}{(z - .72637)(z - .82366 - j.23195)(z - .82366 + j.23195)}$$

$$= \frac{K(z^3 - 3z^2 + 3z - 1)}{z^3 - 2.3737z^2 + 1.9288z - .5319} \quad (25a)$$

with

$$K = 1/G(-1) = .72929 \quad (25b)$$

The theoretical frequency response of $G_{HP}(z)$ in (25), computed as $|G_{HP}(e^{j\omega T})|$ versus ω/ω_s is shown in Fig.8.

The recursive equation for the digital highpass filter follows from (25) as

$$y(k) = 2.3737y(k-1) - 1.9288y(k-2) + .5319y(k-3) + .72929[u(k) - 3u(k-1) + 3u(k-2) - u(k-3)] \quad (26)$$

where $y(k)$ and $u(k)$ are respectively the output and input sequences of the filter. The microprocessor-based implementation and experimental frequency response of the highpass filter (25) or (26) are described in Sections 3 and 4.

$$|G_{HP}(e^{j\omega_d T})|$$

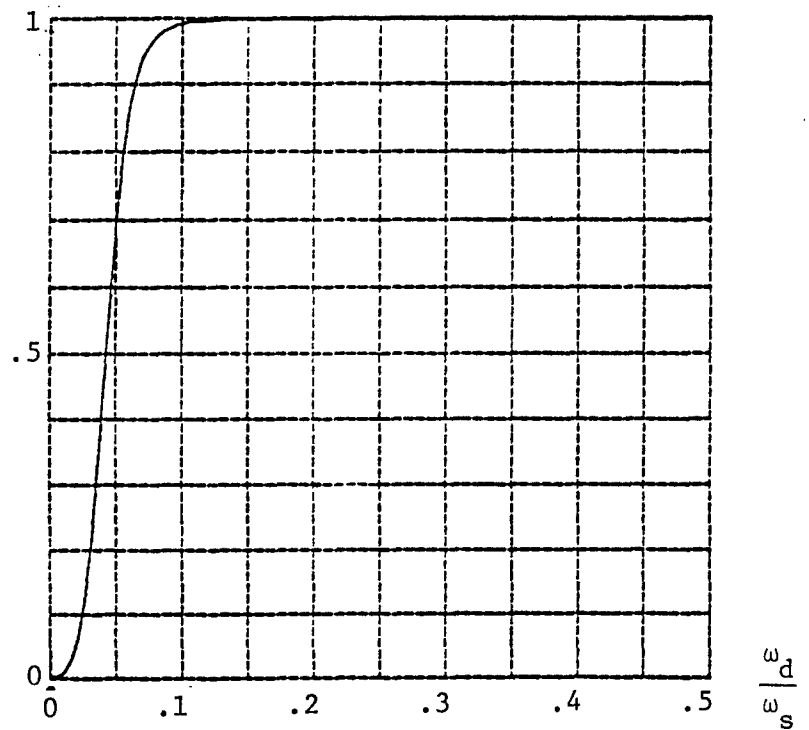


Fig. 8. Theoretical Frequency Response of $G_{HP}(z)$ given by (25).

5.1.4 DESIGN OF DIGITAL BANDPASS FILTERS

The design of a $2n$ -th order analog Butterworth bandpass filter $G_{BP}(s)$ with bandwidth BW and mid-band frequency ω_{ao} is given in Appendix A5. From $G_{BP}(s)$, a corresponding digital Butterworth bandpass filter $G_{BP}(z)$ can be obtained as follows:

- (a) Choose ω_{do}/ω_s , where ω_{do} is the mid-band frequency of the digital filter, so that $(\omega_{do} + \frac{BW}{2})/\omega_s \ll .5$.

- (b) Obtain

$$\begin{aligned} R &= \tan\left(\frac{\omega_{do}}{\omega_s} \pi\right) \\ &= \frac{\omega_{ao} \tau}{2} \end{aligned} \quad (27a)$$

or

$$\tau = \frac{2R}{\omega_{ao}} \quad (27b)$$

- (c) Invoking the conversion scheme, a corresponding $2n$ -th order digital Butterworth bandpass filter can be obtained as^{1,2}

$$\begin{aligned} G_{BP}(z) &= G_{BP}(s) \Big|_{s = \frac{2}{\tau} \frac{z-1}{z+1}} \\ &= \frac{K(z-1)^n (z+1)^n}{(z-p_1)(z-p_1^*) \dots (z-p_n)(z-p_n^*)} \end{aligned} \quad (28a)$$

¹ * denotes complex conjugation.

² Details of manipulation is given in Appendix C.

with

$$p_i \triangleq \frac{(2/\tau + c_i)}{(2/\tau - c_i)}, \quad i = 1, \dots, n, \quad (28b)$$

$$c_i \triangleq \frac{BW}{2} u_i \pm j\omega_{ao}, \quad (28c)$$

$$K \triangleq \frac{(BW)^n (2/\tau)^n}{\prod_{i=1}^n (2/\tau - c_i)(2/\tau - c_i^*)}, \quad (28d)$$

where τ is given by (27).

Remark 3: The design of the bandpass filter (28) assumes that $|BWu_i|^2 \ll 4\omega_{ao}^2$ (see Appendix A5). This is equivalent to considering a bandpass filter with a high Q-factor (the ratio of the midband frequency to the bandwidth), i.e.,

$$Q \triangleq \frac{\omega_{ao}}{BW} \geq 1. \quad (29)$$

Δ

Example 3: 6th Order Digital Butterworth Bandpass Filter

Problem: Design a 6th order digital Butterworth bandpass filter with bandwidth of $BW = 2\pi \times 20$ rad/s and midband frequency of $\omega_{do} = 2\pi \times 20$ rad/s. (Note the Q-factor = $\frac{\omega_o}{BW} = 1$.)

Solution: From Appendix A5, a 6th order analog Butterworth bandpass filter having the above specification ($BW = 2\pi \times 20$, $\omega_{ao} = \omega_{do} = 2\pi \times 20$) is given by

$$G_{BP}(s) = \frac{(BW)^3 s^3}{(s - c_1)(s - c_1^*)(s - c_2)(s - c_2^*)(s - c_3)(s - c_3^*)} \quad (30a)$$

with

$$\begin{aligned} c_1, c_1^* &= \frac{(2\pi \times 20)(-1)}{2} \pm j2\pi \times 20 \\ &= -20\pi \pm j40\pi \end{aligned} \quad (30b)$$

$$\begin{aligned} c_2, c_3 &= \frac{(2\pi \times 20)}{2} (-.5 + j.866) \pm j2\pi \times 20 \\ &= -10\pi + j57.32\pi, \quad -10\pi - j22.68\pi \end{aligned} \quad (30c)$$

$$c_2^*, c_3^* = -10\pi - j57.32\pi, \quad -10\pi + j22.68\pi \quad (30d)$$

- (a) Select $\omega_{do}/\omega_s = .2$ or $\omega_s = 5\omega_{do}$. With this ratio, the relative position of the passband with respect to the sampling frequency is approximately as shown in Fig. 9.

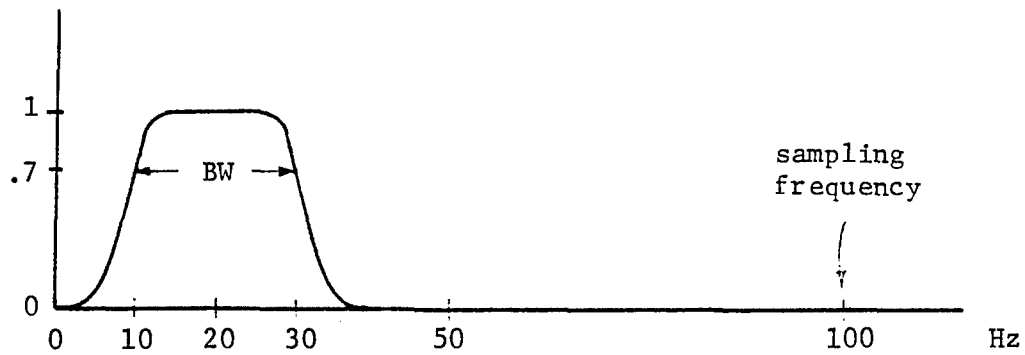


Fig. 9. BandPass Filter

$$(b) \quad R = \tan(.2\pi) \approx .73$$

$$\frac{2}{\tau} = 1.172 \quad (31)$$

$$(c) \quad P_1, P_1^* = \frac{172 + (-20\pi \pm j40\pi)}{172 - (-20\pi \pm j40\pi)} = .6246e^{+j1.3476}$$

$$= .1382 \pm j.6091 \approx .14 \pm j.61 \quad (32a)$$

$$P_2, P_2^* = \frac{172 + (-10\pi \pm j57.32\pi)}{172 - (-10\pi \pm j57.32\pi)} = .8409e^{+j1.6326}$$

$$= -.0519 \pm j.8393 \approx -.052 \pm j.84 \quad (32b)$$

$$P_3, P_3^* = \frac{172 + (-10\pi \pm j22.68\pi)}{172 - (-10\pi \pm j22.68\pi)} = .7319e^{+j.8194}$$

$$= .4996 + j.5348 \approx .50 + j.53 \quad (32c)$$

$$\left(\frac{2}{\tau} - c_1\right)\left(\frac{2}{\tau} - c_1^*\right) = 266.35^2$$

$$\left(\frac{2}{\tau} - c_2\right)\left(\frac{2}{\tau} - c_2^*\right) = 271.68^2$$

$$\left(\frac{2}{\tau} - c_3\right)\left(\frac{2}{\tau} - c_3^*\right) = 215.54^2$$

$$K = \frac{(2\pi \times 20)^3 (172)^3}{266.35^2 \times 271.68^2 \times 215.54^2} = .0415 \quad (32d)$$

Hence,

$$\begin{aligned} G_{BP}(z) &\approx \frac{0.0415(z-1)^3(z+1)^3}{(z-.14-j.61)(z-.14+j.61)(z+.052-j.84)(z+.052+j.84)} \\ &\times \frac{1}{(z-.5-j.53)(z-.5+j.53)} \\ &= \frac{0.0415(z^6 - 3z^4 + 3z^2 - 1)}{(z^6 - 1.176z^5 + 1.7778z^4 - 1.3219z^3 + 1.0035z^2 - .3611z + .1473)} \end{aligned} \quad (33)$$

The theoretical frequency response of $G_{BP}(z)$ in (33), computed as $|G_{BP}(e^{j\omega T})|$ versus ω/ω_s , is shown in Fig. 10.

The recursive equation for the 6th order digital bandpass filter follows from (33) as

$$y(k) = 1.176y(k-1) - 1.7778y(k-2) + 1.3219y(k-3) - 1.0035y(k-4)$$

$$\begin{aligned} & + .3611y(k-5) - .1473y(k-6) + .0415u(k) - .1245u(k-2) + .1245u(k-4) \\ & - .0415u(k-6) \end{aligned} \tag{34}$$

where $y(k)$ and $u(k)$ are respectively the output and input of the filter. The microprocessor-based implementation and the experimental frequency response of the bandpass filter (33) or (34) are given in Sections 3 and 4.

$$|G_{BP}(e^{j\omega_d T})|$$

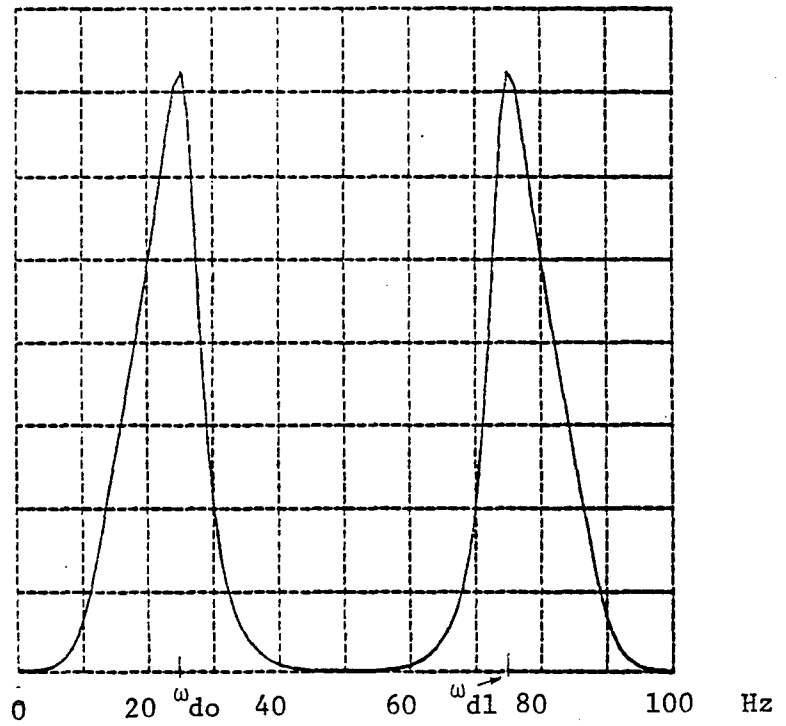


Fig. 10. Theoretical Frequency Response of $G_{BP}(z)$ given by (33).

5.1.5 DESIGN OF DIGITAL BANDSTOP FILTERS

The design of a $2n$ -th order Butterworth bandstop filter $G_{BS}(s)$ with bandwidth BW and midband frequency ω_{ao} is given in Appendix A6. Using $G_{BS}(s)$, a corresponding digital Butterworth bandstop filter $G_{BP}(z)$ can be obtained as follows:

- (a) Choose ω_{do}/ω_s , where ω_{do} is the desired midband frequency of the digital filter, so that $(\omega_{do} + \frac{BW}{2})/\omega_s \ll .5$.
- (b) Obtain

$$R = \tan \left(\frac{\omega_{do}}{\omega_s} \pi \right) = \frac{\omega_{ao} \tau}{2} \quad , \quad (35a)$$

so that

$$\tau = \frac{2R}{\omega_{ao}} \quad . \quad (35b)$$

- (c) Using the bilinear transformation, a corresponding $2n$ -th order digital Butterworth bandstop filter can be obtained as

$$G_{BS}(z) = G_{BS}(s) \quad \left| \quad s = \frac{2}{\tau} \frac{z-1}{z+1} \right.$$

$$= \frac{K(z - z_o)^n (z - z_o^{-1})^n}{(z - p_1)(z - p_1^*) \dots (z - p_n)(z - p_n^*)} \quad , \quad (36a)$$

where

$$p_i \triangleq \frac{2/\tau + c_i}{2/\tau - c_i} \quad , \quad (36b)$$

$$c_i \triangleq \frac{BW}{2} u_i \pm j\omega_{ao} \quad , \quad (\text{see Appendix A5}) \quad (36c)$$

$$K \triangleq \frac{[(2/\tau)^2 + \omega_{ao}^2]^n}{(2/\tau - c_1)(2/\tau - c_1^*) \dots (2/\tau - c_n)(2/\tau - c_n^*)}, \quad (36e)$$

$$z_o = (2/\tau - j\omega_{ao}) / (2/\tau + j\omega_{ao}). \quad (36f)$$

Example 4: 6th Order Digital Butterworth Bandstop Filter

Problem: Design a 6th order digital Butterworth bandstop filter with
 BW = $2\pi \times 20$ rad/s and midband frequency of $\omega_{do} = 2\pi \times 20$ rad/s.
 (Note the Q-factor = $\omega_{do}/BW = 1$.)

Solution: From Appendix A6, a 6th order analog Butterworth bandstop
 filter having the above specification (BW = $2\pi \times 20$ and
 $\omega_{ao} = \omega_{do} = 2\pi \times 20$) is given by

$$G_{BS}(s) = \frac{(s^2 + \omega_{ao}^2)^3}{(s - c_1)(s - c_1^*)(s - c_2)(s - c_2^*)(s - c_3)(s - c_3^*)} \quad (37a)$$

$$c_1, c_1^* = -20\pi \pm j40\pi \quad (37b)$$

$$c_2, c_2^* = -10\pi \pm j57.32\pi \quad (37c)$$

$$c_3, c_3^* = -10\pi \pm j22.68\pi \quad (37d)$$

We note that c_i in (37) are the same as those of (30) due to certain
 similarities in the design specification of Examples 3 and 4. For
 convenience, let us also choose the design variables as in Example 3.
 That is:

- (a) $\omega_{do}/\omega_s = .2$. With this ratio, the relative position of the stopband
 with respect to the sampling frequency is approximately as shown in
 Fig. 11 ,

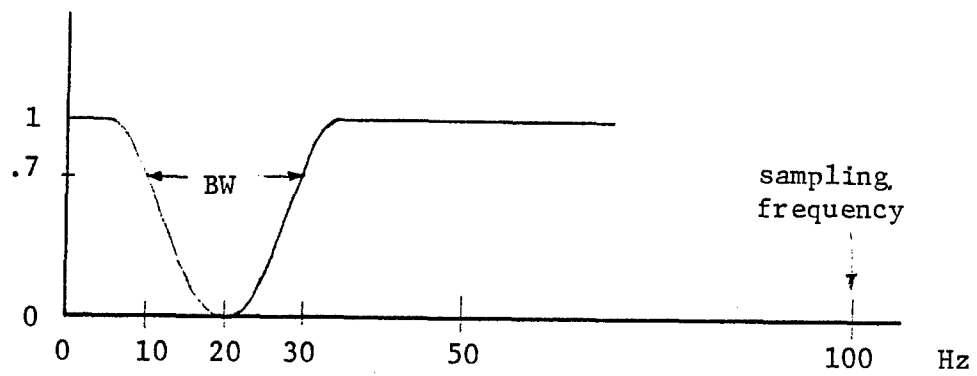


Fig. 11. Bandstop Filter

(b) $R = \tan (.2\pi) \approx .73 ,$

$$2/\tau = 172 ; \quad (38)$$

(c) $p_1, p_1^* = .14 \pm j.61,$

$$p_2, p_2^* = -.052 \pm j.84 , \quad (39)$$

$$p_3, p_3^* = .50 \pm j.53 ,$$

$$\begin{aligned}
\left(\frac{2}{T} - c_1\right)\left(\frac{2}{T} - c_1^*\right) &= 266.35^2, \\
\left(\frac{2}{T} - c_2\right)\left(\frac{2}{T} - c_2^*\right) &= 271.68^2, \\
\left(\frac{2}{T} - c_3\right)\left(\frac{2}{T} - c_3^*\right) &= 215.54^2.
\end{aligned} \tag{40}$$

In addition, we compute

$$\left(\frac{2}{T}\right)^2 + \omega_{ao}^2 = 172^2 + (2\pi \times 20)^2 = 45375 \tag{41a}$$

so that (36e) yields

$$K = \frac{45375^3}{266.35^2 \times 271.68^2 \times 215.54^2} = .384. \tag{41b}$$

The zeroes specified by (36d) are given by

$$z_o = \frac{172 + j2\pi \times 20}{172 - j2\pi \times 20} = e^{-j1.2619} = .3040 - j.9527 \tag{42a}$$

$$z_o^{-1} = .3040 + j.9527. \tag{42b}$$

From the above, the corresponding 6th order digital Butterworth bandstop filter is obtained as

$$\begin{aligned}
G_{BS}(z) &= \frac{.384(z - .3040 + j.9527)^3(z - .3040 - j.9527)^3}{(z - .14 - j.61)(z - .14 + j.61)(z + .052 - j.84)(z + .052 + j.84)} \\
&\times \frac{1}{(z - .5 - j.53)(z - .5 + j.53)}
\end{aligned}$$

$$= \frac{.384(z^6 - 1.824z^5 + 4.109z^4 - 3.873z^3 + 4.109z^2 - 1.824z + 1)}{z^6 - 1.176z^5 + 1.7778z^4 - 1.3219z^3 + 1.0035z^2 - .3611z + .1473} \quad (43)$$

The theoretical frequency response of $G_{BS}(z)$ in (43), computed as $|G_{BS}(e^{j\omega T})|$ versus ω/ω_s is shown in Fig.12.

The recursive equation for the 6th order Butterworth bandstop filter follows from (43) as

$$\begin{aligned} y(k) = & 1.176y(k-1) - 1.7778y(k-2) + 1.3219y(k-3) - 1.0035y(k-4) \\ & + .3611y(k-5) - .1473y(k-6) + .384[u(k) - 1.824u(k-1) + 4.109u(k-2) \\ & - 3.873u(k-3) + 4.109u(k-4) - 1.824u(k-5) + u(k-6)], \end{aligned} \quad (44)$$

where $y(k)$ and $u(k)$ are respectively the output and input of the digital filter. The microprocessor-based implementation and the experimental frequency response of the bandstop filter (43) or (44) are given in Sections 3 and 4.

$$|G_{BS}(e^{j\omega_d T})|$$

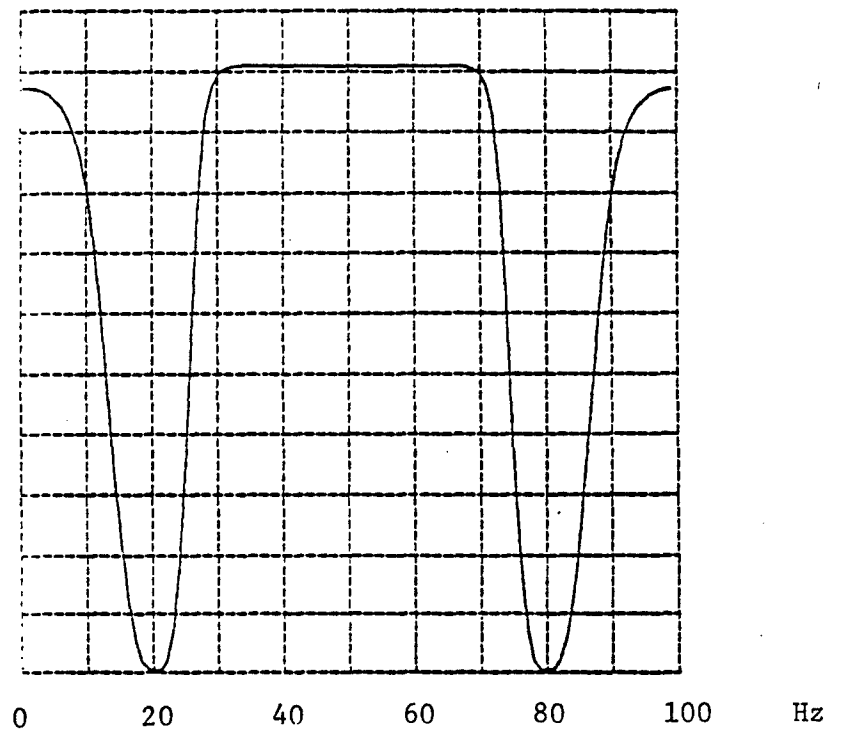


Fig. 12. Theoretical Frequency Response of $G_{BS}(z)$ given by (43).

5.1.6 A FURTHER EXAMPLE

The effect of pole variation in the filter may be vividly illustrated by considering the following example.

Example 5: 3rd Order Digital Chebychev Lowpass Filter

Problem: Design a 3rd order digital Chebychev lowpass filter with cut-off frequency ω_{dc} , having a ripple factor $r = 1$ db. Similar design considerations for Example 1 may be used.

Solution: Let a 3rd order analog Chebychev lowpass filter be denoted by

$$G_{LPC}(s) = \frac{K_a}{(s - u'_1)(s - u'_2)(s - u'_3)} \quad (45)$$

where the poles u'_1 , u'_2 and u'_3 may be determined as follows.

The specification of the filter translates into the frequency response sketched in Fig. 13. The ripple factor

$$\begin{aligned} r &= 1 \\ &= 20 \log(1) - 20 \log\left(\frac{1}{\sqrt{1 + \epsilon^2}}\right) \\ &= 10 \log(1 + \epsilon^2), \end{aligned} \quad (46a)$$

so that

$$\begin{aligned} \epsilon &= \sqrt{10^{r/10} - 1} \\ &= .5088, (r = 1) \end{aligned} \quad (46b)$$

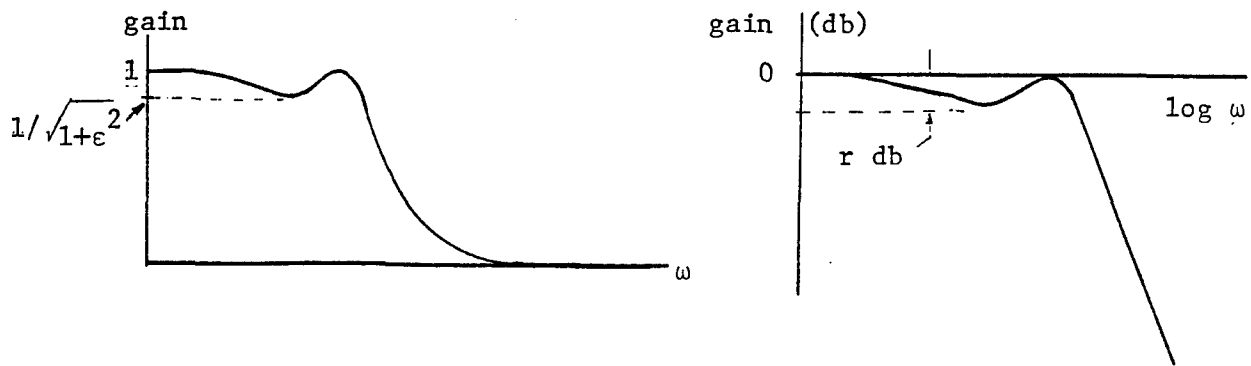


Fig. 13. Chebychev Lowpass Filter

Let $n = 3$ denote the order of the filter and define

$$\begin{aligned}
 a &= \frac{1}{n} \sinh \left(\frac{1}{\epsilon} \right) \\
 &= \frac{1}{n} \ln \left(\frac{1}{\epsilon} + \sqrt{\left(\frac{1}{\epsilon} \right)^2 + 1} \right) \\
 &= \frac{1}{3} \ln \left(\frac{1}{.5088} + \sqrt{\left(\frac{1}{.5088} \right)^2 + 1} \right) \\
 &= .4760 ,
 \end{aligned} \tag{47a}$$

so that

$$\begin{aligned}
 \tanh a &= \frac{e^a - e^{-a}}{e^a + e^{-a}} \\
 &= .4430 .
 \end{aligned} \tag{47b}$$

The poles of the analog Chebychev lowpass filter may be treated as being the poles of the analog Butterworth lowpass filter whose real parts are reduced by a factor of $\tanh a$. Using the values of u_1 in Example 1, the poles of the "normalized" analog Cheybechev filter (45) can hence be determined from

$$u'_i = \operatorname{Re}\{u_i\} \times \tanh a + j\operatorname{Im}\{u_i\}. \quad (48)$$

as

$$u'_1 = -.443 \quad (49a)$$

$$u'_2 = -.2215 + j.866 \quad (49b)$$

$$u'_3 = -.2215 - j.866 \quad (49c)$$

Using relationship (14b), the poles u'_i in the s-domain can be mapped into the poles in the z-domain as

$$p'_i = \frac{(1 + u'_i R)}{(1 - u'_i R)} \quad (50)$$

With the same choice of R as in Example 1 (i.e., $R = .5$), we obtain

$$p'_1 = \frac{1 + (-.443)(.5)}{1 - (-.443)(.5)} = .644 \quad (51a)$$

$$p'_2 = \frac{1 + (-.2215 + j.866)(.5)}{1 - (-.2215 + j.866)(.5)} = .8269e^{j.8247} = .563 + j.609 \quad (51b)$$

$$p'_3 = p_2'^* = .563 - j.609 \quad (51c)$$

Hence, a corresponding 3rd order digital Chebychev lowpass filter can be obtained as

$$G_{\text{LPC}}(z) = \frac{K(z + 1)^3}{(z - p'_1)(z - p'_2)(z - p'_3)} \quad (52)$$

$$\begin{aligned}
&= \frac{K(z+1)^3}{(z - .644)(z - .563 - j.609)(z - .563 + j.609)} \\
&= \frac{K(z^3 + 3z^2 + 3z + 1)}{(z^3 - 1.77z^2 + 1.413z - .443)} \\
&\triangleq KG(z) \quad , \quad (52a)
\end{aligned}$$

where for unity gain at the low frequency

$$K = \frac{1}{G(1)} = \frac{.2}{8} = .025 \quad . \quad (52b)$$

The theoretical frequency response of $G_{LPC}(z)$ given by (52), computed as $|G_{LPC}(e^{j\omega T})|$ versus ω/ω_s is shown in Fig. 14.

The recursive equation for the 3rd order Chebychev lowpass filter follows from (52) as

$$\begin{aligned}
y(k) = & 1.77y(k-1) - 1.413y(k-2) + .443y(k-3) \\
& + .025u(k) + .075u(k-1) + .075u(k-2) + .025u(k-3) \quad , \quad (53)
\end{aligned}$$

where $y(k)$ and $u(k)$ are the output and the input of the digital filter. The microprocessor-based implementation and experimental frequency response of the digital Chebychev lowpass filter (52) or (53) are given in Sections 3 and 4.

$$|G_{LPC}(e^{j\omega_d T})|$$

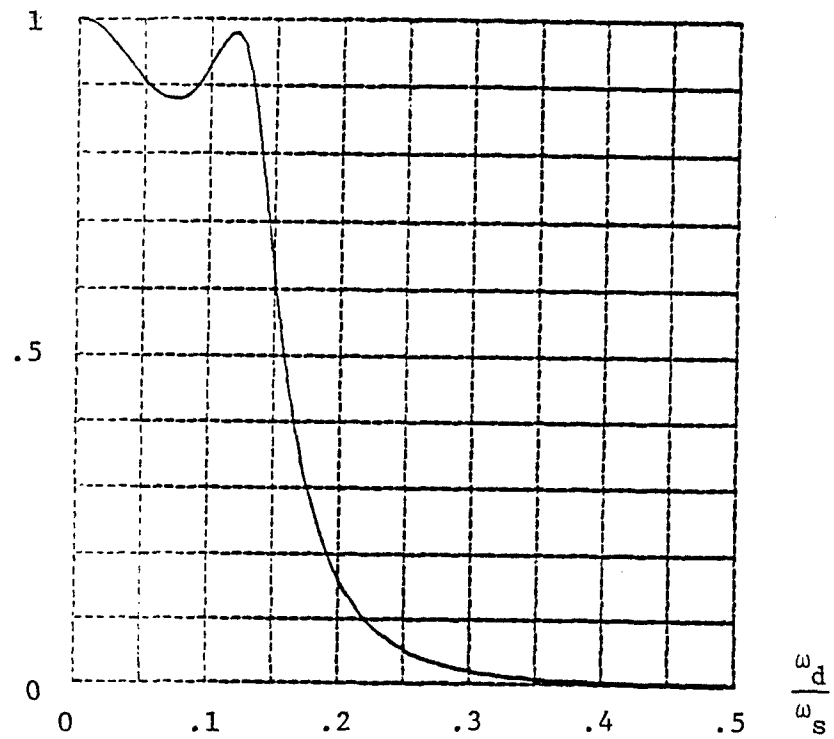


Fig. 14. Theoretical Frequency Response of $G_{LPC}(z)$ given by (52).

5.2 MICROPROCESSOR REALIZATION OF DIGITAL FILTERS

The microprocessor firmware for implementing the digital frequency selective filters designed in Section 2 is described below.

Fig.15 shows the block diagrams of the hardware for the microprocessor-based signal processing system used in the realization of the digital filters. The CPU of the system is the Motorola MC6802 microprocessor, which is supported by 2716 EPROM (monitor program), 6810 PAM's, 6821 PIA's, keypad, L.E.D. display and interfacing buffers to form the microcomputer called MOUSE¹ (see Fig.16). The microcomputer operates at a clock rate of 1 MHz.

The data acquisition unit is the Datel MDAS-16 multiple (multiplex) channel 12-bit A/D converter with a conversion time of about 20 μ s per data. A Datel Hzl2BGC 12 bit D/A converter with a settling time of 3 μ s is used as a zero-order-hold output of the microprocessor-based system. The interface between the 8-bit MC6802 and the 12-bit I/O (A/D and D/A) peripherals are done through 6821 PIA's (Fig.17). The software (< 1k bytes) for the digital filters are stored in the external 2716 EPROM. The additional 2114 RAM's provide handy facilities for debugging and immediate alterations of the software if desired. The wiring diagrams for the microprocessor-based signal processing system is shown in Figs. 16 and 17.

¹ Acronym for Microcomputer of Oakland University School of Engineering.

The main consideration in the microprocessor software for the digital filters involves the development of fast and efficient arithmetic subroutines, the scaling of the recursive digital filter equations, the handling of saturation and some memory management.

A fast 3 bytes x 1 byte multiplication subroutine with an execution time of about 96 μ s was developed for the digital filter implementation. Other main subroutines include a 3 bytes + 3 bytes summation (about 85 μ s), transfer and negation of 3 byte data. Details of these subroutines are given in Appendix B.

In order to minimize the occurrence of saturation (overflow or underflow) in the finite wordlength data (3 bytes or 24 bits), the recursive formulas for each of the filters will be scaled in a fashion similar to those done in an analog computer simulation. The scaled recursive equations for the digital filters from Examples 1 - 5, are shown in Table 1.

It is remarked that the forward gains K of the filters may be reduced, if necessary, to achieve a proper scaling which will not saturate the filter output. Alternatively, the output saturation can be handled by use of overflow test instructions in the microprocessor software.

The 6802 microprocessor software for implementing the digital frequency selective filters described in Table 1 are given in Appendix B.

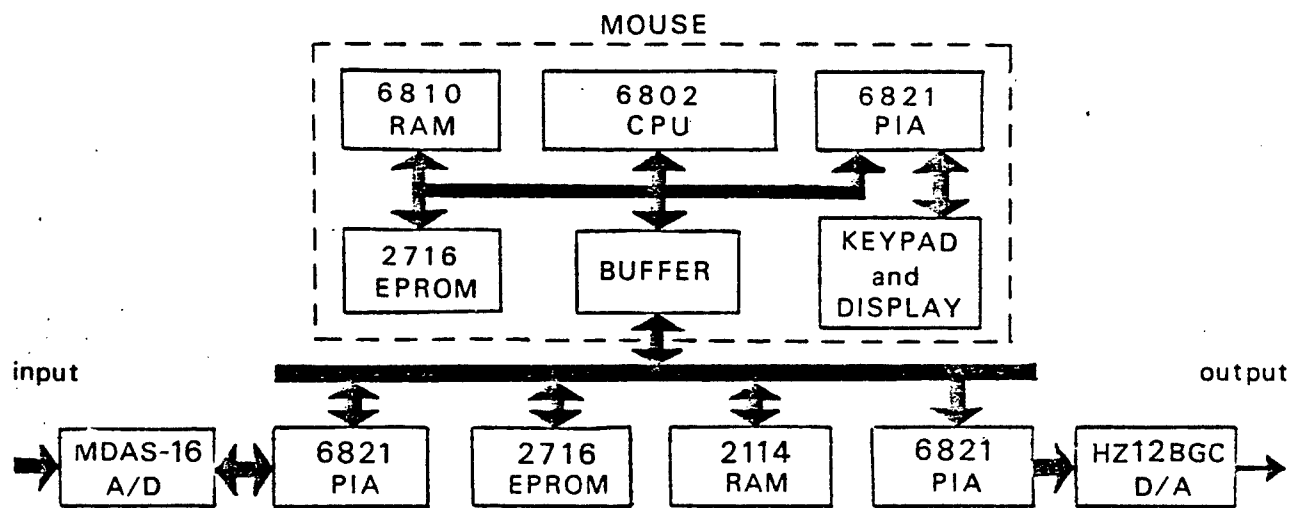
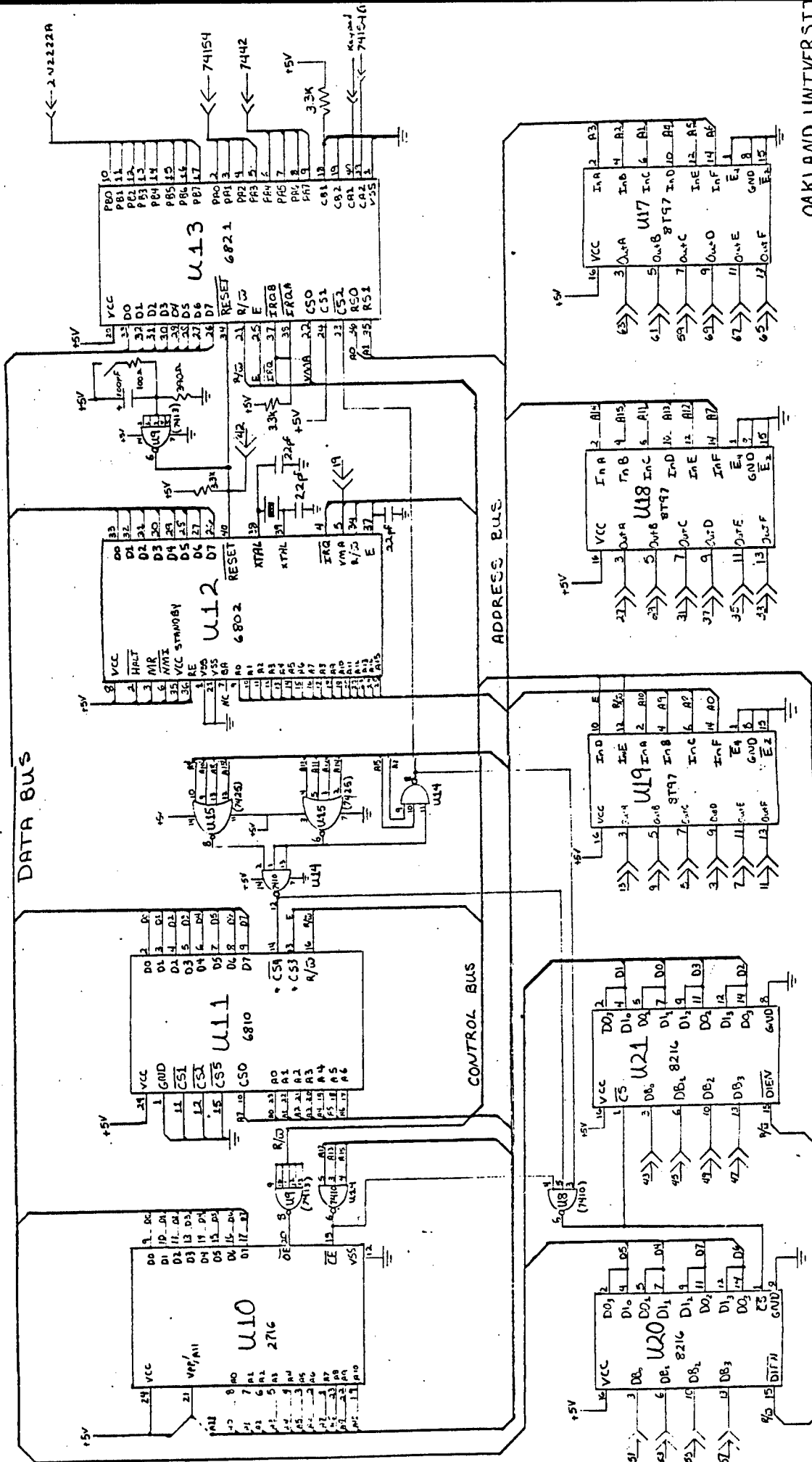


Fig. 15 Block Diagram of Microprocessor System



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SCHEMATIC

Fig. 16a Schematic for MOUSE (reference[9])

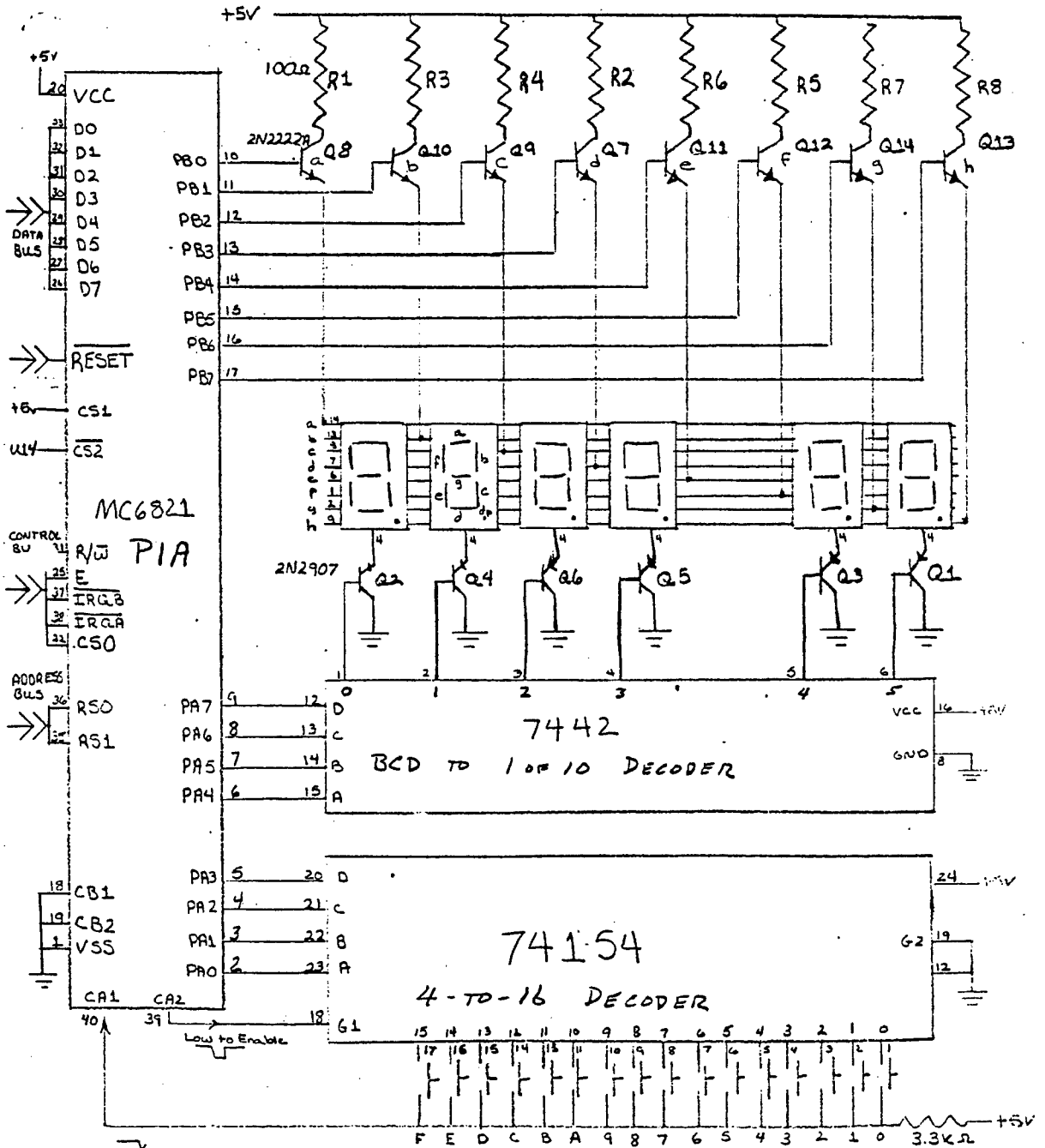


Fig.16b LED Display and Keypad Schematic
(reference[9])

OAKLAND UNIVERSITY
OU-2
DIGITAL DISPLAY

1	VSS	CA1	40
2	PA0	CA2	39
3	PA1	IRQA	38
4	PA2	IRQB	37
5	PA3	RSD	36
6	PA4	RSI	35
7	PA5	MC6821 RESET	34
8	PA6	00	33
9	PA7	01	32
10	PB0	02	31
11	PB1	03	30
12	PB2	04	29
13	PB3	05	28
14	PB4	06	27
15	PB5	07	26
16	PB6	E	25
17	PB7	CS1	24
18	CB1	CS2	23
19	CB2	CS0	22
20	VCC	R/W	21

1	VSS	RESET	40
2	WALT	XTAL	39
3	MR	XTAL	38
4	IRQA	E	37
5	VMA	RE	36
6	NMI	STANDBY VCC	35
7	BA	MC6802 R/W	34
8	VCC	00	33
9	A0	01	32
10	A1	02	31
11	A2	03	30
12	A3	04	29
13	A4	05	28
14	A5	06	27
15	A6	07	26
16	A7	A15	25
17	A8	A16	24
18	A9	A13	23
19	A10	A12	22
20	A11	VSS	21

1	A7	VCC	24
2	A6	A8	23
3	A5	A9	22
4	A4	VPP/RA1	21
5	A3	OE	20
6	A2	A10	19
7	A1	CE	18
8	A0	D7	17
9	D0	D6	16
10	D1	D5	15
11	D2	D4	14
12	VSS	D3	13

2716 (2732)

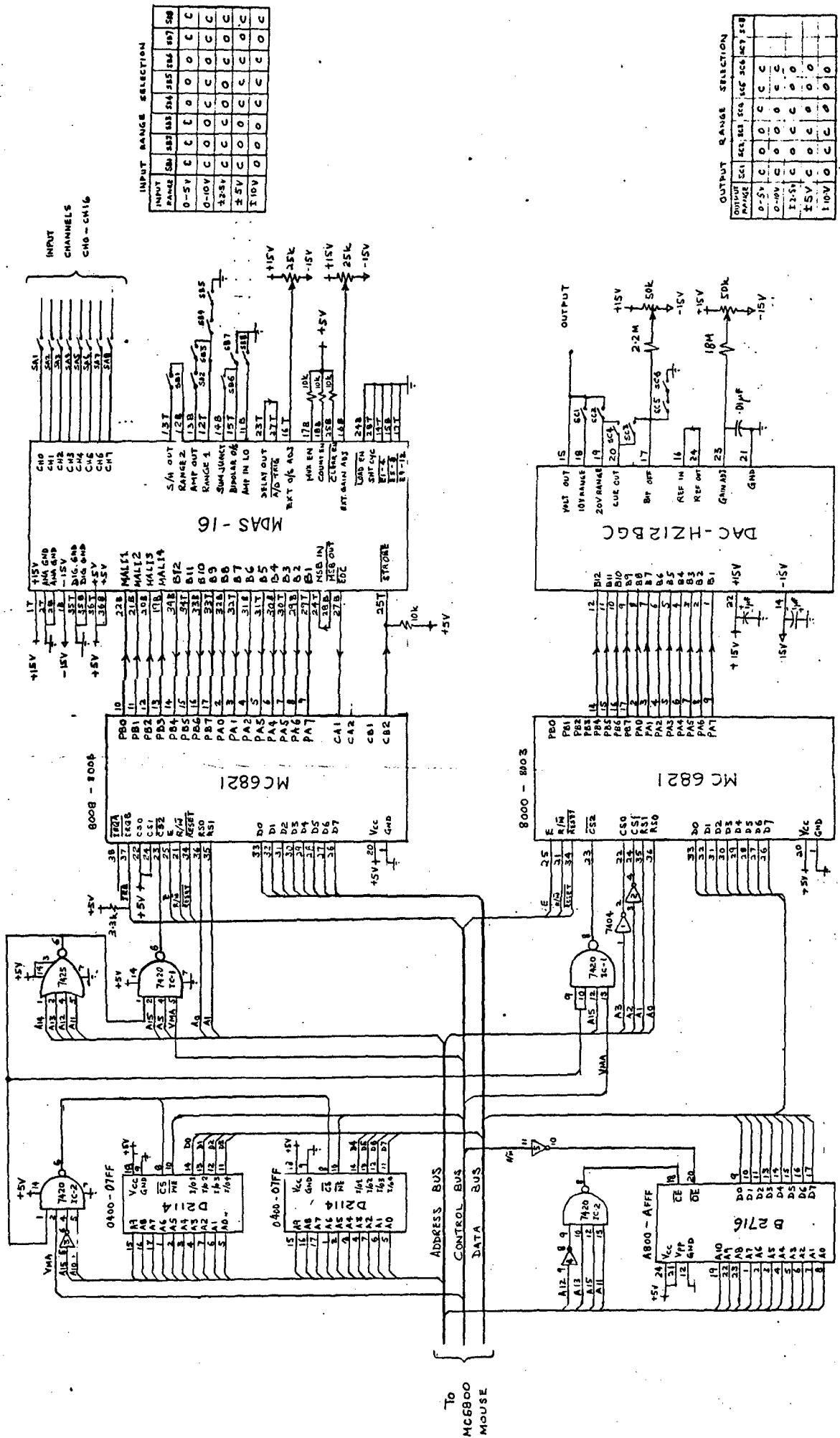
1	GND	VCC	24
2	D0	A0	23
3	D1	A1	22
4	D2	A2	21
5	D3	A3	20
6	D4	A4	19
7	D5	A5	18
8	D6	A6	17
9	D7	R/W	16
10	CS0	CS5	15
11	CS1	CS4	14
12	CS2	CS3	13

MC6810

1	0	VCC	24
2	1	A	23
3	2	B	22
4	3	C	21
5	4	D	20
6	5	G2	19
7	6	G1	18
8	7	15	17
9	8	H	16
10	9	13	15
11	10	12	14
12	GND	11	13

74154

Fig. 16c Pinouts of major chips on MOUSE
(reference [9])



INPUT RANGE SELECTION

INPUT RANGE	5A	5B	5C	5D	5E	5F	5G	5H	5I	5J
0-5V	C	C	C	C	C	C	C	C	C	C
0-10V	C	C	C	C	C	C	C	C	C	C
±2.5V	C	C	C	C	C	C	C	C	C	C
±5V	C	C	C	C	C	C	C	C	C	C
±10V	C	C	C	C	C	C	C	C	C	C

OUTPUT RANGE SELECTION

OUTPUT RANGE	5C1	5C2	5C3	5C4	5C5	5C6	5C7	5C8
0-5V	C	C	C	C	C	C	C	C
0-10V	C	C	C	C	C	C	C	C
±2.5V	C	C	C	C	C	C	C	C
±5V	C	C	C	C	C	C	C	C
±10V	C	C	C	C	C	C	C	C

Fig. 17 MICROPROCESSOR CONTROLLED ADC & DAC

Table 1: SCALED RECURSIVE EQUATIONS FOR THE DIGITAL FILTERS

Butterworth Low-Pass (Example 1)

$$.84y(k) = y(k-1) - .6y(k-2) + .12y(k-3) \\ + .04u(k) + .12u(k-1) + .12u(k-2) + .04u(k-3)$$

Butterworth High-Pass (Example 2)

$$.4213y(k) = y(k-1) - .8126y(k-2) + .2241y(k-3) \\ + .3072u(k) - .9217u(k-1) + .9217u(k-2) - .3072u(k-3)$$

Butterworth Band-Pass (Example 3)

$$.5625y(k) = .6615y(k-1) - y(k-2) + .7436y(k-3) - .5645y(k-4) + .2031y(k-5) \\ - .0829y(k-6) + .0233u(k) - .0700u(k-2) + .0700u(k-4) \\ - .0233u(k-6)$$

Butterworth Band-Stop (Example 4)

$$.5625y(k) = .6615y(k-1) - y(k-2) + .7436y(k-3) - .5645y(k-4) + .2031y(k-5) \\ - .0829y(k-6) + .216u(k) - .394u(k-1) + .8875u(k-2) \\ - .8366u(k-3) + .8875u(k-4) - .394u(k-5) + .216u(k-6)$$

Chebyshev Low-Pass (Example 5)

$$.565y(k) = y(k-1) - .7983y(k-2) + .2503y(k-3) \\ + .0141u(k) + .0423u(k-1) + .0423u(k-2) + .0141u(k-3)$$

5.3 EXPERIMENTAL FREQUENCY RESPONSE OF MICROPROCESSOR-BASED DIGITAL FILTERS

The experimental set-up for recording the frequency response of the microprocessor-based digital filters is shown in Fig. 18. The test input $u(t)$, generated by the voltage controlled oscillator, consists of a constant amplitude sinusoidal signal whose frequency modulates or sweeps (sufficiently) slowly from low frequency to high frequency and vice versa, i.e.,

$$u(t) = A \sin \omega t$$

where the frequency ω is controlled by a triangular or saw-tooth signal $w(t)$. The digital output of the filter is recorded on a storage scope whose horizontal axis is driven by the same $w(t)$. From the set-up, one can experimentally determine the frequency responses of the microprocessor-based digital filters. Figs. 19-23 show the actual experimental frequency responses of the microprocessor-based 3rd order filters designed in the examples of Section 2.

For comparison, the critical frequencies of the theoretical filters and the implemented microprocessor-based filters are tabulated in Table 2. As shown in the table, the specification of the filters in terms of ω_{dc} , ω_{do} and BW have been met satisfactorily.

It is important to note that the critical frequencies ω_{dc} and ω_{do} can readily be altered by simply adjusting the sampling frequency ω_s . There is, however, an upper bound on the maximum possible sampling frequency which can be used for the filter due to the finite speed of the microprocessor. Nevertheless, the design specifications concerned in the present investigation can be satisfactorily fulfilled by the current generation of 8-bit microprocessors. For more stringent design specifications, one may resort to the new generation of 16-bit microprocessors and/or use of high speed arithmetic logic chips.

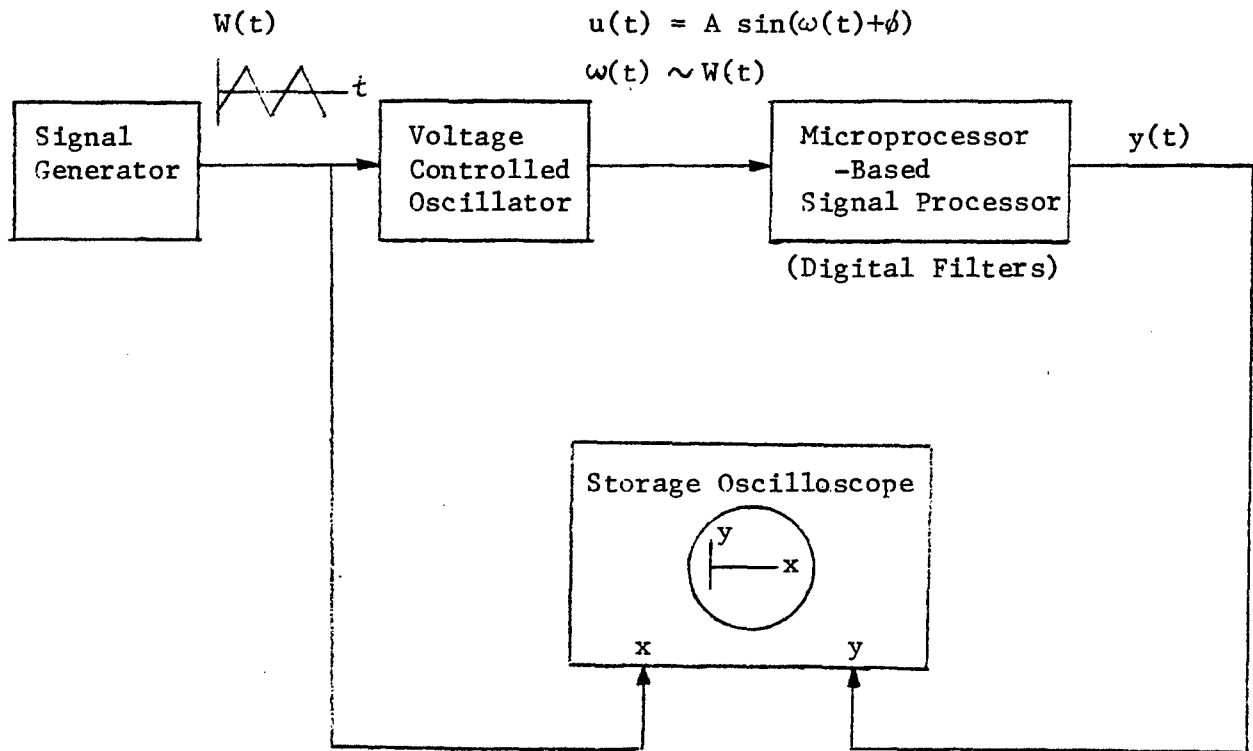


Fig. 18 Experimental Set-up for Measurement of Frequency Response

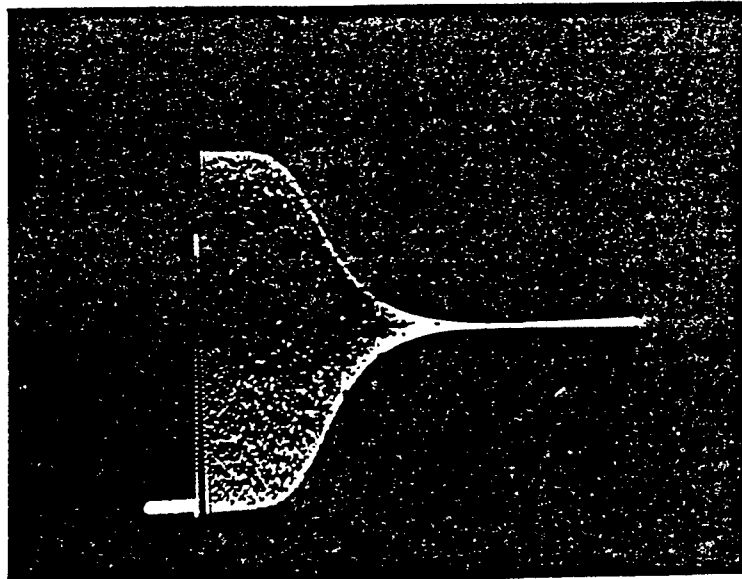


Fig.19 Experimental Frequency Response of Microprocessor-Based 3rd Order Butterworth Lowpass Filter $G_{LP}(z)$, (Example 1).

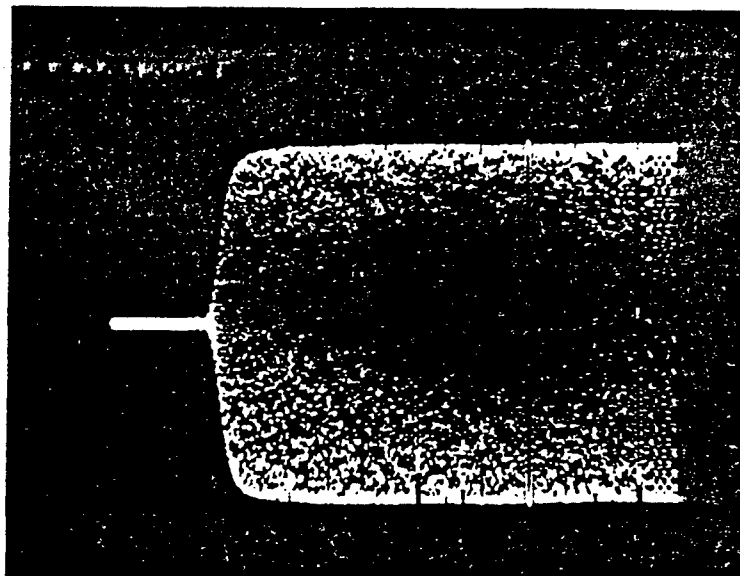


Fig.20 Experimental Frequency Response of Microprocessor-Based 3rd Order Butterworth Highpass Filter $G_{HP}(z)$ (Example 2)

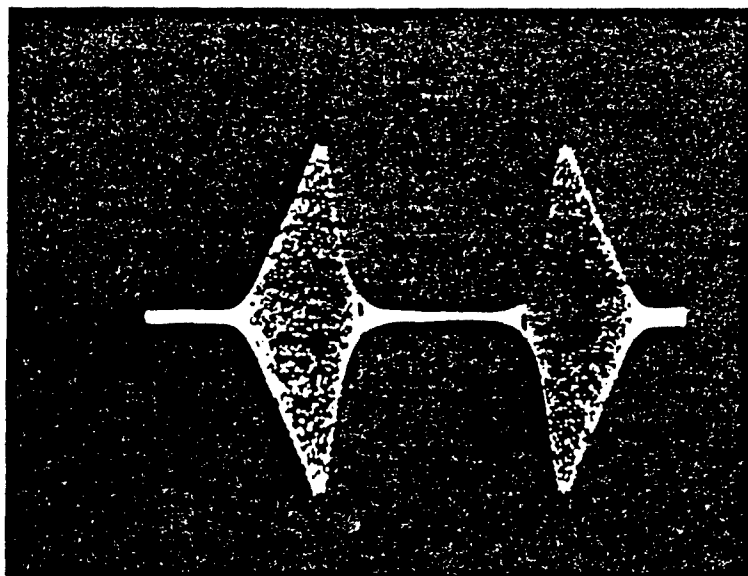


Fig.21 Experimental Frequency Response of Microprocessor-Based
6th Order Butterworth Bandpass Filter $G_{BP}(z)$ (Example 3)

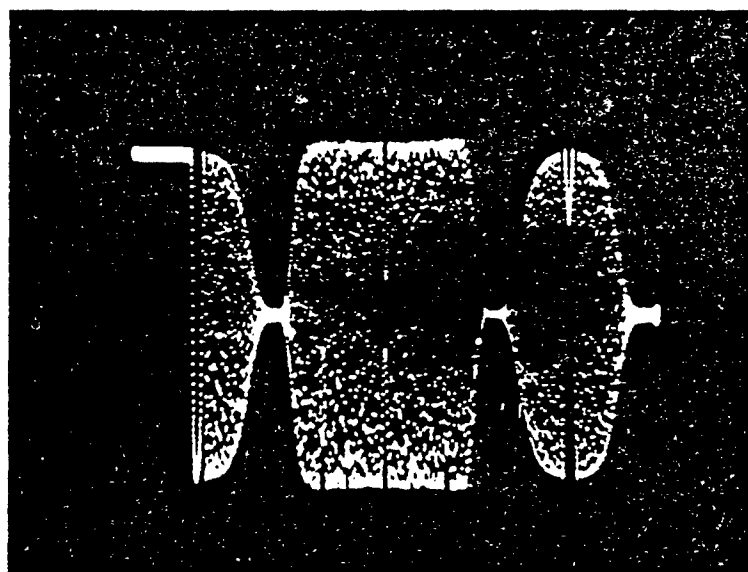


Fig.22 Experimental Frequency Response of Microprocessor-Based
6th Order Butterworth Bandstop Filter $G_{BS}(z)$ (Example 4)

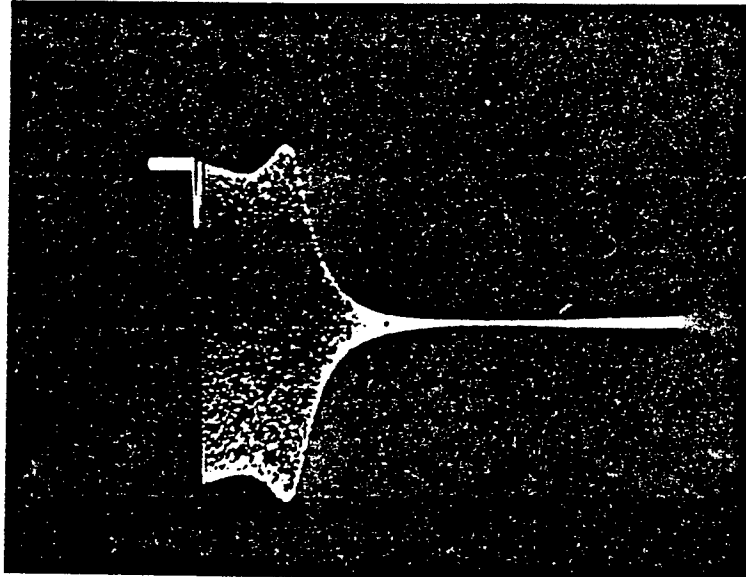


Fig.23 Experimental Frequency Response of Microprocessor-Based
3rd Order Chebychev Lowpass Filter (Example 5)

Table 2. Critical Frequencies of Theoretical and Microprocessor-Based Filters

Digital Filter		Sampling Frequency ω_s (Hz)	Cut-Off or Cut-In Frequency ω_{dc} (Hz)	Frequency at Attenuation = 0.1 $\omega_{0.1}$ (Hz)	Remarks	
3rd Order Butterworth Lowpass	Theoretical	200	30	52		
	Experimental	200	27	49		
3rd Order Chebychev Lowpass	Theoretical	200	29	44	The cut-off rate is faster than that of the Butterworth filter.	
	Experimental	200	27	45		
3rd Order Butterworth Highpass	Theoretical	200	10	5		
	Experimental	200	9.5	5		
Digital Filter		Sampling Frequency ω_s (Hz)	Midband Frequency ω_o (Hz)	First Aliasing Midband Freq. ω_1 (Hz)	Bandwidth BW (Hz)	Remarks
6th Order Butterworth Bandpass	Theoretical	100	25	75	7.5	$\omega_o/\omega_s = .25$
	Experimental	140	28	100	17	$\omega_o/\omega_s = .2$
6th Order Butterworth Bandstop	Theoretical	100	20	80	14	
	Experimental	104	16	78	16	

REFERENCES

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- [9] Haskell R.E. and G.A. Jackson, "Designing with microprocessors hardware and software," School of Engineering, Oakland University, Rochester, MI 48063, Feb., 1980.

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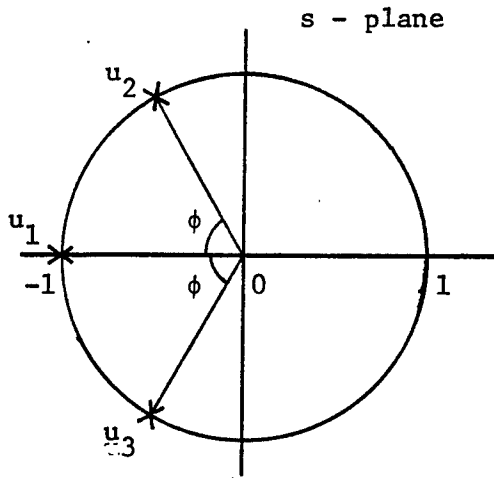
APPENDIX A

SUMMARY OF ANALOG BUTTERWORTH FREQUENCY SELECTIVE FILTERS

A1. Low-Pass Filter $G_{LPN}(s)$ with Normalized Cut-Off Frequency ($\omega_{ac} = 1$):

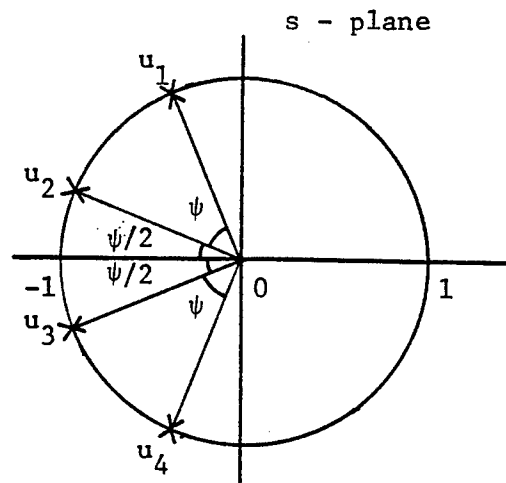
$$G_{LPN}(s) = \frac{1}{(s - u_1)} \frac{1}{(s - u_2)} \cdots \frac{1}{(s - u_n)} \quad (A1)$$

where¹ u_i , $i = 1, \dots, n$, are the stable poles which lie on the unit circle in the s -plane as shown in Fig.A1.



$$\phi = \frac{\pi}{n}$$

$n = \text{odd}$ (e.g., $n=3$)



$$\psi = \frac{\pi}{n}$$

$n = \text{even}$ (e.g., $n=4$)

Fig. A1. Pole Locations of Normalized Butterworth Lowpass Filter

¹ We note that complex poles must occur in complex conjugate pairs in order for the filter to be physically realizable.

A2: Low-Pass Filter $G_{LP}(s)$ with Arbitrary Cut-Off Frequency ω_{ac} :

To translate $G_{LPN}(s)$ into $G_{LP}(s)$, one substitutes s/ω_{ac} for s in (A.1) and obtain

$$G_{LP}(s) = \frac{\omega_{ac}}{(s - \omega_{ac} u_1)} \frac{\omega_{ac}}{(s - \omega_{ac} u_2)} \dots \frac{\omega_{ac}}{(s - \omega_{ac} u_n)} \quad (A.2)$$

A typical set of pole locations of $G_{LP}(s)$ is shown in Fig.A2.

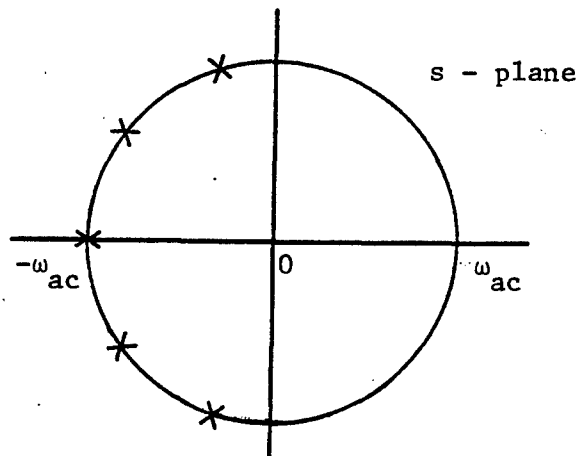


Fig. A2. Pole Locations of Butterworth Lowpass Filter with ω_{ac} ($n = 5$)

A3: High-Pass Filter $G_{HPN}(s)$ with Normalized Cut-In Frequency ($\omega_{ac} = 1$):

To translate $G_{LPN}(s)$ into $G_{HPN}(s)$, one substitutes $1/s$ for s in (A.1) and obtains

$$\begin{aligned}
 G_{HPN}(s) &= \frac{s}{(1 - su_1)} \frac{s}{(1 - su_2)} \cdots \frac{s}{(1 - su_n)} \\
 &= \frac{(-1/u_1)s}{(s - 1/u_1)} \frac{(-1/u_2)s}{(s - 1/u_2)} \cdots \frac{(-1/u_n)s}{(s - 1/u_n)} \\
 &= \frac{s}{(s - u_1)} \frac{s}{(s - u_2)} \cdots \frac{s}{(s - u_n)}, \tag{A.3}
 \end{aligned}$$

where the last equality follows from the fact that $1/u_i = u_i^*$, $|u_i| = 1$ and u_i occur in complex conjugate pairs. A typical set of poles and zeroes for $G_{HPN}(s)$ is shown in Fig.A3.

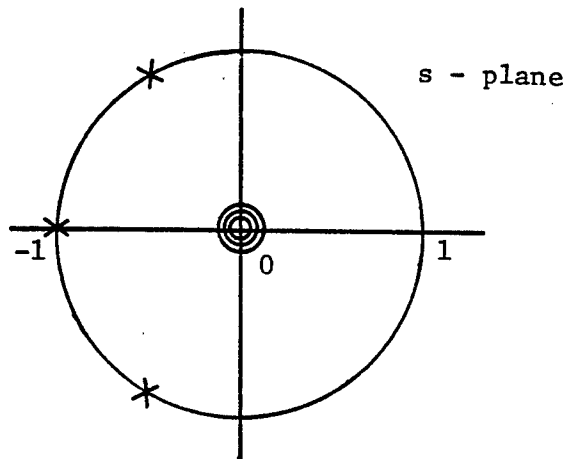


Fig. A3. Pole-Zero locations of Normalized Butterworth Highpass Filter. ($n = 3$)

A4: High-Pass Filter $G_{HP}(s)$ with Arbitrary Cut-In Frequency ω_{ac} :

To translate $G_{HPN}(s)$ to $G_{HP}(s)$, one substitutes s/ω_{ac} for s in (A.3) and obtains

$$G_{HP}(s) = \frac{s}{(s - \omega_{ac} u_1)} \frac{s}{(s - \omega_{ac} u_2)} \cdots \frac{s}{(s - \omega_{ac} u_n)},$$

where a typical set of poles and zeroes for $G_{HP}(s)$ is shown in Fig.A4.

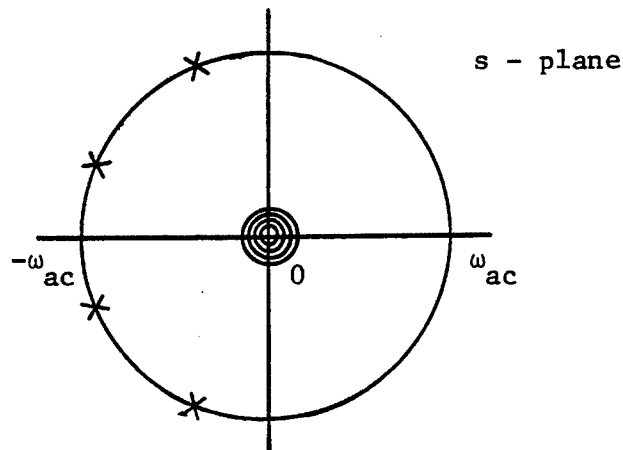


Fig. A4. Pole-Zero Locations of Butterworth Highpass Filter with ω_{ac} . ($n = 4$)

A5: Band-Pass Filter $G_{BP}(s)$ with Bandwidth BW and Midband Frequency ω_{ao} :

A high Q-factor will generally be assumed, i.e., $Q \triangleq \omega_{ao}/BW \geq 1$.
 To translate $G_{LPN}(s)$ to $G_{BP}(s)$, one substitutes

$$\frac{1}{BW} \frac{s^2 + \omega_{ao}^2}{s} \quad \text{for } s \text{ in (A.1) and obtains}$$

$$\begin{aligned} G_{BP}(s) &= \frac{BW \cdot s}{(s^2 - 2BWu_1s + \omega_{ao}^2)} \cdot \frac{BW \cdot s}{(s^2 - 2BWu_2s + \omega_{ao}^2)} \cdots \frac{BW \cdot s}{(s^2 - 2BWu_ns + \omega_{ao}^2)} \\ &\triangleq \frac{BW \cdot s}{(s - p_1)(s - q_1)} \cdot \frac{BW \cdot s}{(s - p_2)(s - q_2)} \cdots \frac{BW \cdot s}{(s - p_n)(s - q_n)} \\ &\approx \frac{BW \cdot s}{(s - c_1)(s - c_1^*)} \cdot \frac{BW \cdot s}{(s - c_2)(s - c_2^*)} \cdots \frac{BW \cdot s}{(s - c_n)(s - c_n^*)}, \quad (\text{A.5}) \end{aligned}$$

where

$$p_i, q_i \triangleq \frac{BW}{2} u_i \pm \frac{1}{2} \sqrt{(BWu_i)^2 - 4\omega_{ao}^2}$$

which may be approximated by

$$c_i, c_i^* \triangleq \frac{BW}{2} u_i \pm j\omega_{ao}$$

for a high Q-factor. A typical pole-zero location of $G_{BP}(s)$ is shown in Fig. A5.

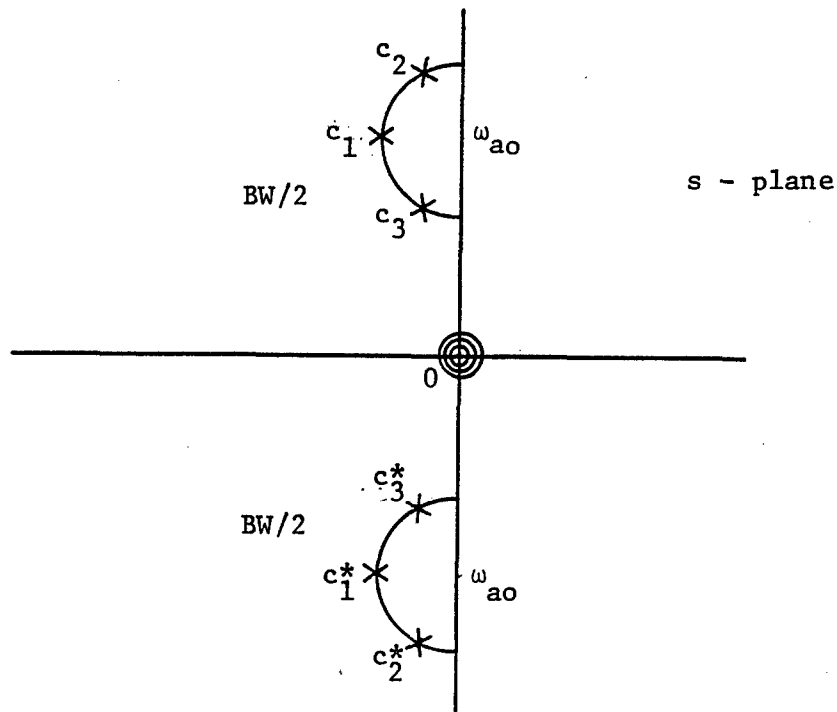


Fig. A5. Pole-Zero Locations of Butterworth Bandpass Filter with Bandwidth BW and Midband Frequency ω_{ao} .

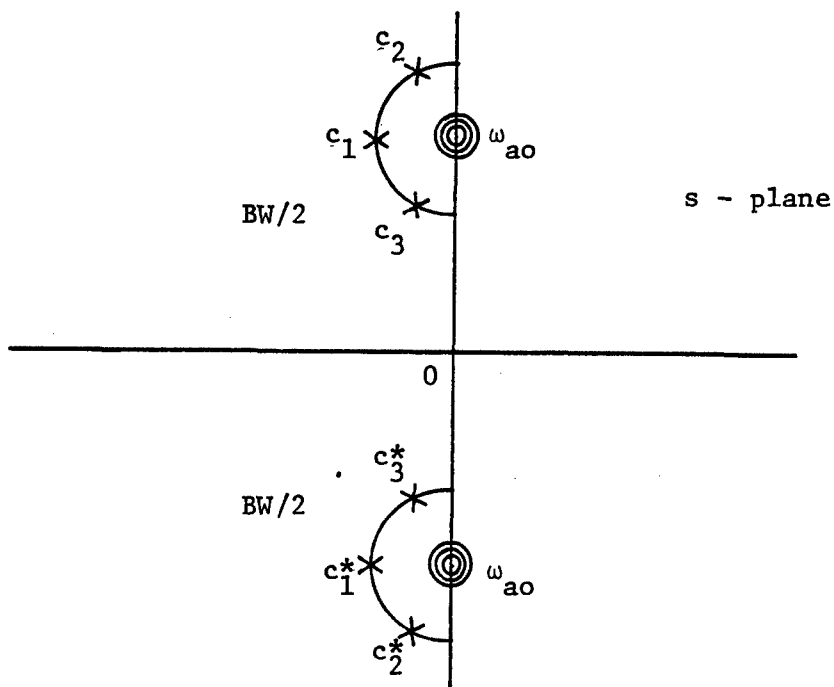


Fig. A6. Pole-Zero Locations of Butterworth Bandstop Filter with Bandwidth BW and Midband Frequency ω_{ao} .

A6: Band-Stop Filter $G_{BS}(s)$ with Bandwidth BW and Midband Frequency ω_{ao} :

A high Q-factor will similarly be assumed, i.e., $Q \triangleq \omega_{ao}/BW \geq 1$.

To translate $G_{HPN}(s)$ to $G_{BS}(s)$, one substitutes

$\frac{s}{s^2 + \omega_{ao}^2}$ for s in (A.3) and obtains

$$G_{BS}(s) = \frac{(s^2 + \omega_{ao}^2)}{(s^2 - BWu_1s + \omega_{ao}^2)} \frac{(s^2 + \omega_{ao}^2)}{(s^2 - BWu_2s + \omega_{ao}^2)} \cdots \frac{(s^2 + \omega_{ao}^2)}{(s^2 - BWu_n s + \omega_{ao}^2)}$$

$$\approx \frac{(s+j\omega_{ao})(s-j\omega_{ao})}{(s - c_1)(s - c_1^*)} \frac{(s+j\omega_{ao})(s-j\omega_{ao})}{(s - c_2)(s - c_2^*)} \cdots \frac{(s+j\omega_{ao})(s-j\omega_{ao})}{(s - c_n)(s - c_n^*)}$$

(A.6)

where c_i are as defined for (A5). A typical pole-zero location for $G_{BS}(s)$ is shown in Fig. A6.

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APPENDIX B

MICROPROCESSOR PROGRAMS FOR DIGITAL FREQUENCY SELECTIVE FILTERS

MICROKIT 6800 DISK ASSEMBLER

PAGE 1
VER 1

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*
*
0008 IDLTIM EQU X'08'
000A INITAD EQU X'0A'
000C DESTIN EQU X'0C'
000E XTEMP EQU X'0E'
0019 CHNO EQU X'19'
*
0000 ORG X'A800'
A800 DEA900 MOVEBLK LDX #X'A900'
A803 DF00 STX X'00'
A805 DE0400 LDX #X'0400'
A808 DF0C STX DESTIN
A80A DE00 LDX X'00'
A80C DF0A STX INITAD
A80E DE0A MOVEBYT LDX INITAD
A810 A600 LDAA X'00',X
A812 08 INX
A813 DF0A STX INITAD
A815 DE0C LDX DESTIN
A817 A700 STAA X'00',X
A819 0C0/FF CPX #X'07FF'
A81C 2705 BEQ BEQ6
A81E 08 INX
A81F DF0C STX DESTIN
A821 20E8 BRA MOVEBYT
A823 3F BEQ6
*

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```

A824 7F0015 MULT1 CLR X'15'
A827 7F0016 CLR X'16'
A82A 7F0017 CLR X'17'
A82D CE0008 LDX #X'0008'
A830 D610 LDAB X'10'
A832 770011 BNE11 ASR X'11'
A835 760012 ROR X'12'
A838 760013 ROR X'13'
A83B 58 ASLB
A83C 2412 BCC BCC11
A83E 9613 LDAA X'13'
A840 9B17 ADDA X'17'
A842 9717 STAA X'17'
A844 9612 LDAA X'12'
A846 9916 ADCA X'16'
A848 9716 STAA X'16'
A84A 9611 LDAA X'11'
A84C 9915 ADCA X'15'
A84E 9715 STAA X'15'
A850 09 BCC11 DEX
A851 26DF BNE ENE11
A853 39 RTS

*
A854 A630 MUL LDAA X'30' ,X
A856 9710 STAA X'10' ,X
A858 A600 LDAA X'00' ,X
A85A 9711 STAA X'11' ,X
A85C A610 LDAA X'10' ,X
A85E 9712 STAA X'12' ,X
A860 A620 LDAA X'20' ,X
A862 9713 STAA X'13' ,X
A864 DF0E STX XTEMP
A866 BDAB24 JSR MULT1
A869 DE0E LDX XTEMP
A86B 9615 LDAA X'15'
A86D A708 STAA X'08' ,X
A86F 9616 LDAA X'16' ,X
A871 A718 STAA X'18' ,X
A873 9617 LDAA X'17' ,X
A875 A728 STAA X'20' ,X
A877 9C0C CPX DESTIN
A879 2703 BEQ OUT1
A87B 08 INX
A87C 20D6 BRA MUL
A87E 39 OUT1 RTS

*
A87F 7F002F SUM CLR X'2F'
A882 7F003F CLR X'3F'
A885 7F004F CLR X'4F'
A888 09 DEX
A889 08 BNE10 INX
A88A 964F LDAA X'4F'
A88C AB20 ADDA X'20' ,X
A88E 974F STAA X'4F'
A890 963F LDAA X'3F'
A892 A910 ADCA X'10' ,X
A894 973F STAA X'3F'
    
```

```

A896 962F      LDAA X'2F'
A898 A900      ADCA X'00',X
A89A 2Z2F      STAA X'2F'
A89C 9C0C      CPX  DESTIN
A89E 26E9      BNE  BNE10
A8A0 39        RTS
  
```

```

*
A8A1 6020      NEGATE NEG  X'20',X
A8A3 8600      LDAA #X'00'
A8A5 A210      SECA X'10',X
A8A7 A710      STAA X'10',X
A8A9 8600      LDAA #X'00'
A8AB A200      SECA X'00',X
A8AD A700      STAA X'00',X
A8AF 39        RTS
  
```

```

*
A8B0 962F      TRANSF LDAA X'2F'
A8B2 A700      STAA X'00',X
A8B4 963F      LDAA X'3F'
A8B6 A710      STAA X'10',X
A8B8 964F      LDAA X'4F'
A8BA A720      STAA X'20',X
A8BC 39        RTS
  
```

```

*
A8BD 7F8001     PIASU1 CLR  X'8001'
A8C0 7F8003     CLR  X'8003'
A8C3 7F8009     CLR  X'8009'
A8C6 7F800B     CLR  X'800B'
A8C9 86FF      LDAA #X'FF'
A8CB B78000     STAA X'8000'
A8CE B78002     STAA X'8002'
A8D1 #F        CLRA
A8D2 B78008     STAA X'8008'
A8D5 860F      LDAA #X'0F'
A8D7 B7800A     STAA X'800A'
A8DA 8634      LDAA #X'34'
A8DC B78001     STAA X'8001'
A8DF B78003     STAA X'8003'
A8E2 B78009     STAA X'8009'
A8E5 862C      LDAA #X'2C'
A8E7 B7800B     STAA X'800B'
A8EA 39        RTS
  
```

```

*
A8EB 9710      MULT2 STAA X'10'
A8ED 7F0011     CLR  X'11'
A8F0 0712      STAB X'12'
A8F2 7F0013     CLR  X'13'
A8F5 BDA824     JSR  MULT1
A8FB 39        RTS
  
```

```

*
A8F9 DE08      IDLE  LDX  IDLTIM
A8FB 09        IDL1  DEX
A8FC 26FD      BNE  IDL1
A8FE 39        RTS
  
```

*

MICROKIT 6800 DISK ASSEMBLER

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```

A8FF                                ORG X'A900'
A900 B607 HILO1 LDAA #X'07'
A902 B7B00A STAA X'800A'
A905 CE002E LDX #X'002E'
A908 BDA8B0 JSR TRANSF
A90E B68008 LDAA X'8008'
A90E 9725 STAA X'25'
A910 F6800A LDAB X'800A'
A913 D735 STAB X'35'

                                *2
A915 CE0025 LDX #X'0025'
A918 DF0C STX DESTIN
A91A BDA854 JSR MUL

                                *3
A919 CE002E LDX #X'002E'
A920 DF0C STX DESTIN
A922 CE0028 LDX #X'0028'
A925 BDAB7F JSR SUM

                                *4
A928 962E LDAA X'2E'
A92A 9722 STAA X'22'
A92C 963E LDAA X'3E'
A92E 9732 STAA X'32'

                                *5
A930 CE0026 LDX #X'0026'
A933 BDAB80 JSR TRANSF
A936 DF0C STX DESTIN
A938 BDA854 JSR MUL

                                *6
A93B CE002D LDX #X'002D'
A93E BDA8B0 JSR TRANSF

                                *
A941 39 RTS

                                *
                                *
                                *
                                *8
A942 962F HILO2 LDAA X'2F'
A944 D63F LDAB X'3F'
A946 087F ECRA #X'7F'
A948 C8F0 EORB #X'E0'
A94A B7B000 STAA X'8000'
A94D F7B002 STAB X'8002'

                                *9
A950 CE0000 LDX #X'0000'
A953 A621 UPYU LDAA X'21',X
A955 E631 LDAB X'31',X
A957 A720 STAA X'20',X
A959 E730 STAB X'30',X
A95B 08 INX
A95C 8C0005 CPX #X'0005'
A95F 26F2 BNE UPYU

                                *10
A961 CE0024 LDX #X'0024'
A964 DF0C STX DESTIN
A966 CE0020 LDX #X'0020'
A969 BDA854 JSR MUL

```

```

                                x11
A96C CE0029 LDX #X'0029'
A96F EDABA1 JSR NEGATE
                                *
A972 39 RTS
                                *
                                *
                                *
A973 CE0010 CLRTIM LDX #X'0010'
A976 6F1F CCCCCC CLR X'1F',X
A97B 6F2F CLR X'2F',X
A97A 6F3F CLR X'3F',X
A97C 6F5F CLR X'5F',X
A97E 6F6F CLR X'6F',X
A980 6FFF CLR X'7F',X
A982 09 DEX
A983 26F1 BNE CCCCCC
A985 CE00C1 LDX #X'00C1'
A988 DF08 STX IDLTIM
A98A 09 RTS
```

MICROKIT 6800 DISK ASSEMBLER - VER 1

```
*
*
* LOW PASS FILTER 1
*
*
A98B 8E07FF LOPAS1 LDS #X'07FF'
A98E BDA9AA          JSR  LOOAT1
A991 BDA8BD GOLD    JSR  PIASU1
A994 BDA900 LP      JSR  HILO1
A99Z CE002E          LDX  #X'002E'
A99A DF0C           STX  DESTIN
A99C CE002D          LDX  #X'002D'
A99F BDA87F          JSR  SUM
A9A2 BDA942          JSR  HILO2
A9A5 BDA8F9          JSR  IDLE
A9AB 20EA           BRA  LP
```

*

```
*  
*  
A9AA BDA973 LDDAT1 JSR CLRRTIM  
A9AD 861F LDAA #X'1F'  
A9AF 9750 STAA X'50'  
A9B1 9753 STAA X'53'  
A9B3 9754 STAA X'54'  
A9B5 860A LDAA #X'0A'  
A9B7 9752 STAA X'52'  
A9B9 9755 STAA X'55'  
A9BB 869A LDAA #X'9A'  
A9BD 9751 STAA X'51'  
A9BF 8619 LDAA #X'19'  
A9C1 9756 STAA X'56'  
A9C3 39 RTS  
*
```

*
 *
 *

CHEBYCHEV LOWPASS FILTER

A9C4 8E07FF	LDS #X'07FF'
A9C7 863E	LDAA #X'3E'
A9C9 9750	STAA X'50'
A9CB 86CC	LDAA #X'CC'
A9CD 9751	STAA X'51'
A9CF 8610	LDAA #X'10'
A9D1 9752	STAA X'52'
A9D3 9755	STAA X'55'
A9D5 8630	LDAA #X'30'
A9D7 9753	STAA X'53'
A9D9 9754	STAA X'54'
A9DB 86C6	LDAA #X'C6'
A9DD 9756	STAA X'56'
A9DF 2090	BRA GOLD

*

```

*
*
* HIGH PASS FILTER 1
*
A9E1 8E07FF HIPAS1 LDS #X'07FF'
A9E4 BDAACF JSR HIDAT1
A9E7 BDABBD GOHI JSR PIASU1
A9EA BDA930 HP JSR HILO1 STEPS 1 TO 6
A9ED CE002E LDX #X'002E'
A9F0 DF0C STX DESTIN STEP 7
A9F2 CE002C LDX #X'002C'
A9F5 BDAB80 JSR TRANSF
A9FB BDAB7F JSR SUM
A9FE BDA942 JSR HILO2 STEP2 8 TO 11
A9FE CE002A LDX #X'002A'
AA01 BDA8A1 JSR NEGATE
AA04 CE002C LDX #X'002C'
AA07 BDA8A1 JSR NEGATE
AA0A BDABF2 JSR IDLE STEP 12
AA0D 20DE BRA HP

```

*

MICROKIT 5800 DISK ASSEMBLER

PAGE
VER

```
      *  
      *  
AA0F B0A973 HIDAT1 JSR CLRTIM  
AA12 B639          LDAA #X'39'  
AA14 9750          STAA X'50'  
AA16 B6D0          LDAA #X'D0'  
AA18 9751          STAA X'51'  
AA1A B64F          LDAA #X'4F'  
AA1C 9752          STAA X'52'  
AA1E 9753          STAA X'53'  
AA20 B6EC          LDAA #X'EC'  
AA22 9753          STAA X'53'  
AA24 9754          STAA X'54'  
AA26 B660          LDAA #X'60'  
AA28 9756          STAA X'56'  
AA2A 39           RTS
```

```
    *  
    *
```

*
* SUBROUTINES FOR BANDPASS & BANDSTOP
*

AA2B 8607 BPBS1 LDAA #X'07'
 AA2D B780CA STAA X'800A'
 AA30 9624 LDAA X'24'
 AA32 D634 LDAB X'34'
 AA34 972C STAA X'2C'
 AA36 D73C STAB X'3C'
 AA3B 9644 LDAA X'44'
 AA3A 974C STAA X'4C'
 AA3C CE002C LDX #X'002C'
 AA3F BDABA1 JSR NEGATE
 AA42 39 RTS

*
*
*

AA43 DF0C BPBS2 STX DESTIN STEP 3
 AA45 CE0068 LDX #X'0068'
 AA4B BDAB7F JSR SUM
 AA4B CE002E LDX #X'002E'
 AA4E BDABE0 JSR TRANSF

*

AA51 DF0C STX DESTIN STEP 4
 AA53 CE0028 LDX #X'0028'
 AA56 BDAB7F JSR SUM

*

AA59 CE0026 LDX #X'0026' STEP 5
 AA5C BDABE0 JSR TRANSF
 AA5F DF0C STX DESTIN
 AA61 BDAB54 JSR MUL
 AA64 CE002E LDX #X'002E'
 AA67 DF0C STX DESTIN
 AA69 CE002D LDX #X'002D'
 AA6C BDABE0 JSR TRANSF
 AA6F BDAB7F JSR SUM

*

AA72 962F LDAA X'2F' STEP 6
 AA74 D63F LDAB X'3F'
 AA76 88ZF EDRA #X'ZF'
 AA78 C8F0 EORB #X'F0'
 AA7A B78000 STAA X'8000'
 AA7D F78002 STAB X'8002'

*

AA80 CE0000 LDX #X'0000' STEP 7
 AA83 A621 YUP LDAA X'21',X
 AA85 E631 LDAB X'31',X
 AA87 A720 STAA X'20',X
 AA89 E730 STAB X'30',X
 AA8B A641 LDAA X'41',X
 AA8D A740 STAA X'40',X
 AA8F 08 INX
 AA90 6C0005 CPX #X'0005'
 AA93 26EE BNE YUP
 AA95 CE0025 LDX #X'0025'
 AA98 BDABE0 JSR TRANSF

*

STEP 8

```
AA9B CE0023 LDX #X'0023'  
AA9E DF0C STX DESTIN  
AAA0 CE0021 LDX #X'0020'  
AAA3 EDAB54 JSR MUL  
AAA6 CE0025 LDX #X'0025'  
AAA9 DF0C STX DESTIN  
AAAB EDAB54 JSR MUL  
#  
AAAE CE0028 LDX #X'0028'  
AAB1 BDABA1 JSR NEGATE  
AAB4 CE002A LDX #X'002A'  
AAB7 BDABA1 JSR NEGATE  
*  
AABA 89 RTS
```

STEP 9

```

      ***
      **
      *
      * BANDPASS FILTER 1
      *
      ***
AACB 8E07FF EPASS1 LDS  #'07FF'
AAE 8DAB14 JSR  BPDAT1
AAC1 8DAB8D GOBP JSR  PIASU1
      *
      * STEP1
AAC4 8DAA2B EP JSR  BPBS1
AAC7 8680C8 LDAA X'800B'
AACA 9763 STAA X'63'
AACC F6809A LDAB X'800A'
AACF D773 STAB X'73'
      *
      * STEP2
AAD1 CE0063 LDX  #'0063'
AAD4 DF0C STX  DESTIN
AAD6 8DAB54 JSR  MUL
      *
      * STEP3
AAD9 CE0068 LDX  #'0068'
AACDC 8DAA43 JSR  BPBS2
      *
      * STEP10
AADF CE0000 LDX  #'0000'
AAE2 A664 UPUBF LDAA X'64',X
AAE4 A760 STAA X'60',X
AAE6 A661 LDAA X'61',X
AAE8 A764 STAA X'64',X
AAEA E674 LDAB X'74',X
AAEC E770 STAB X'70',X
AAEE E671 LDAB X'71',X
AAF0 E774 STAB X'74',X
AAF2 08 INX
AAF3 8C0003 CPX  #'0003'
AAF6 26EA BNE  UPUBF
      *
      * STEP11
AAF8 CE0062 LDX  #'0062'
AAF8 DF0C STX  DESTIN
AAFD CE0060 LDX  #'0060'
AB00 8DAB54 JSR  MUL
      *
      * STEP 12
AB03 CE0068 LDX  #'0068'
AB06 8DAB81 JSR  NEGATE
AB09 CE006A LDX  #'006A'
AB0C 8DAB81 JSR  NEGATE
      *
      * STEP 13
AB0F 8DABF9 JSR  IDLE
AB12 20B0 BRA  EP
      *

```

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```

*
*
AB14 BDA973 BPDAT1 JSR CLRTRM
AB17 B615 LDAA #X'15'
AB19 9750 STAA X'50'
AB1B B634 LDAA #X'34'
AB1D 9751 STAA X'51'
AB1F B691 LDAA #X'91'
AB21 9752 STAA X'52'
AB23 B6BE LDAA #X'BE'
AB25 9753 STAA X'53'
AB27 B6FF LDAA #X'FF'
AB29 9754 STAA X'54'
AB2B B6A9 LDAA #X'A9'
AB2D 9755 STAA X'55'
AB2F B6C3 LDAA #X'C3'
AB31 9756 STAA X'56'
AB33 B606 LDAA #X'06'
AB35 9790 STAA X'90'
AB37 9793 STAA X'93'
AB39 B612 LDAA #X'12'
AB3B 9791 STAA X'91'
AB3D 9792 STAA X'92'
AB3F 39 RTS

```

```

*
*
*
```

```

***
**
*
* BANDSTOP FILTER 1
*
**
***
AB40 8107FF BSTOP1 LDS  #X'07FF'
AB43 BDA697 JSR  BSDAT1
AB46 BDA8BD GOBS JSR  PIASU1
AB49 BDA82B BS JSR  BPES1
AB4C B6B008 LDAA X'8008'
AB4F 9766 STAA X'66'
AB51 F6B00A LDAB X'800A'
AB54 D776 STAB X'76'
*
AB56 CE0066 LDX  #X'0066' STEP 2
AB59 DF0C STX  DESTIN
AB5B BDA854 JSR  MUL
*
AB5E CE006E LDX  #X'006E' STEP 3
AB61 BDA843 JSR  BPBS2
*
AB64 CE0000 LDX  #X'0000' STEP 10
AB67 A661 LPUBS LDAA X'61',X
AB69 A760 STAA X'60',X
AB6B E671 LDAB X'71',X
AB6D E770 STAB X'70',X
AB6F 08 INX
AB70 8C0006 CPX  #X'0006'
AB73 26F2 BNE  UPUBS
*
AB75 CE0065 LDX  #X'0065' STEP 11
AB78 DF0C STX  DESTIN
AB7A CE0060 LDX  #X'0060'
AB7D BDA854 JSR  MUL
*
AB80 CE0069 LDX  #X'0069' STEP 14
AB83 BDA8A1 JSR  NEGATE
AB86 CE006B LDX  #X'006B'
AB89 BDA8A1 JSR  NEGATE
AB8C CE006D LDX  #X'006D'
AB8F BDA8A1 JSR  NEGATE
*
AB92 BDA8F9 JSR  IDLE STEP 15
*
AB95 20E2 BRA  BS
*

```

```

*
*
AB97 BDA973 BSDAT1 JSR CLRTRM
AB9A B615 LDAA #X'1E'
AB9C 9750 STAA X'50'
AB9E B634 LDAA #X'34'
ABA0 9751 STAA X'51'
ABA2 B691 LDAA #X'91'
ABA4 9752 STAA X'52'
ABA6 B6EE LDAA #X'EE'
ABAB 9753 STAA X'53'
ABAA B6A9 LDAA #X'A9'
ABAC 9755 STAA X'55'
ABAE B6FF LDAA #X'FF'
AB80 9754 STAA X'54'
AB82 B6C3 LDAA #X'C3'
AB84 9756 STAA X'56'
AB86 B637 LDAA #X'37'
AB88 9790 STAA X'90'
AB8A 9796 STAA X'96'
AB8C B665 LDAA #X'65'
AB8E 9791 STAA X'91'
AB90 9795 STAA X'95'
AB92 B6E3 LDAA #X'E3'
AB94 9792 STAA X'92'
AB96 9794 STAA X'94'
AB98 B6D6 LDAA #X'D6'
AB9A 9795 STAA X'93'
AB9C B9 RTS

```

```

**
*
**

```

```

AB9D END MOVBLK

```

MACROKKT

A 0000
 INITAD 000A
 MOVEBK AB00
 BNE11 AB32
 SUM AB7F
 FIAGU1 AB8D
 HILD1 A900
 CCCCCC A976
 LODAT1 A9AA
 HIDAT1 AA0F
 EFASS1 AAEE
 EPDAT1 AB14
 UFUES AB67

AB00

B 0000
 DESTIN 00QC
 MOVEBYT AB0F
 BCC11 AB50
 BNE10 AB89
 MULT2 ABER
 HIL02 A942
 LOPAS1 A98B
 HIPAS1 A9E1
 EFBS1 AA2B
 GOEP AAC1
 ESTOP1 AB40
 BSDAT1 AB97

CPUSK

X 0000
 XTEMP 000E
 BEG6 AB23
 MUL AB54
 NEGATE AB81
 IDLE ABF9
 LPYU A953
 GOLO A991
 COMI A9E7
 BPEB2 AA43
 EF AAC4
 GOES AB46
 CHEBY A9C4

ABEHEMELER

0000
 000E
 AB23
 AB54
 AB81
 ABF9
 A953
 A991
 A9E7
 AA43
 AAC4
 AB46
 A9C4

IDLTIM
 CHND
 MULT1
 OUT1
 TRANSF
 IDL1
 CLRTIM
 LP
 HP
 YUP
 UPUBP
 BS

000B
 0019
 AB24
 AB7E
 AB8D
 ABFB
 A973
 A994
 A9EA
 AA03
 AAEE
 AB49

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APPENDIX C

DERIVATION OF DIGITAL FILTERS

C.1. Derivation of Eq. (14)

From (A.2),

$$G_{LP}(s) = \frac{\omega_{ac}}{(s - \omega_{ac} u_1)} \frac{\omega_{ac}}{(s - \omega_{ac} u_2)} \dots \frac{\omega_{ac}}{(s - \omega_{ac} u_n)}$$

Using the substitution in (14), one obtains

$$\begin{aligned} G_{LP}(z) &= G_{LP}(s) \Big|_{s = \frac{\omega_{ac}}{R} \frac{z-1}{z+1}} \\ &= \frac{\omega_{ac}}{\left(\frac{\omega_{ac}}{R} \frac{z-1}{z+1} - \omega_{ac} u_1\right)} \dots \frac{\omega_{ac}}{\left(\frac{\omega_{ac}}{R} \frac{z-1}{z+1} - \omega_{ac} u_n\right)} \\ &= \frac{R(z+1)}{[z-1 - Ru_1(z+1)]} \dots \frac{R(z+1)}{[z-1 - Ru_n(z+1)]} \\ &= \frac{R^n}{(1-u_1 R) \dots (1-u_n R)} \frac{(z+1)^n}{\left(z - \frac{1+u_1 R}{1-u_1 R}\right) \dots \left(z - \frac{1+u_n R}{1-u_n R}\right)} \\ &\triangleq K \frac{(z+1)^n}{(z-p_1) \dots (z-p_n)} \end{aligned}$$

(c.f. Section 2.2)

C.2. Derivation of Eq. (21)

From (A.3),

$$G_{HP}(s) = \frac{s}{(s - \omega_{ac} u_1)} \frac{s}{(s - \omega_{ac} u_2)} \dots \frac{s}{(s - \omega_{ac} u_n)}$$

Using the substitution in (21), one obtains

$$\begin{aligned}
 G_{HP}(z) &= G_{HP}(s) \Big|_{s = \frac{\omega_{ac}}{R} \frac{z-1}{z+1}} \\
 &= \frac{\frac{\omega_{ac}}{R} \frac{z-1}{z+1}}{\left(\frac{\omega_{ac}}{R} \frac{z-1}{z+1} - \omega_{ac} u_1\right)} \dots \frac{\frac{\omega_{ac}}{R} \frac{z-1}{z+1}}{\left(\frac{\omega_{ac}}{R} \frac{z-1}{z+1} - \omega_{ac} u_n\right)} \\
 &= \frac{z-1}{[z-1 - u_1 R(z+1)]} \dots \frac{z-1}{[z-1 - u_n R(z+1)]} \\
 &= \frac{1}{(1-u_1 R) \dots (1-u_n R)} \frac{(z-1)^n}{\left(z - \frac{1+u_1 R}{1-u_1 R}\right) \dots \left(z - \frac{1+u_n R}{1-u_n R}\right)} \\
 &\triangleq K \frac{(z-1)^n}{(z-p_1) \dots (z-p_n)}
 \end{aligned}$$

(c.f. Section 2.3).

C.3. Derivation of Eq. (28)

From (A.5)

$$G_{BP}(s) = \frac{BW \cdot s}{(s-c_1)(s-c_1^*)} \frac{BW \cdot s}{(s-c_2)(s-c_2^*)} \dots \frac{BW \cdot s}{(s-c_n)(s-c_n^*)}$$

Using the substitution in (28), one obtains

$$\begin{aligned}
 G_{BP}(z) &= G_{BP}(s) \Big|_{s = \frac{2}{T} \frac{z-1}{z+1}} \\
 &= \frac{BW \frac{2}{T} \frac{z-1}{z+1}}{\left(\frac{2}{T} \frac{z-1}{z+1} - c_1\right) \left(\frac{2}{T} \frac{z-1}{z+1} - c_1^*\right)} \dots \frac{BW \frac{2}{T} \frac{z-1}{z+1}}{\left(\frac{2}{T} \frac{z-1}{z+1} - c_n\right) \left(\frac{2}{T} \frac{z-1}{z+1} - c_n^*\right)} \\
 &= \frac{BW \frac{2}{T} (z-1)(z+1)}{\left(\frac{2}{T} - c_1\right) \left(\frac{2}{T} - c_1^*\right) \left[z - \frac{\left(\frac{2}{T} + c_1\right)}{\left(\frac{2}{T} - c_1\right)}\right] \left[z - \frac{\frac{2}{T} - c_1^*}{\left(\frac{2}{T} - c_1^*\right)}\right]} \dots \frac{BW \frac{2}{T} (z-1)(z+1)}{\left(\frac{2}{T} - c_n\right) \left(\frac{2}{T} - c_n^*\right) \left[z - \frac{\left(\frac{2}{T} + c_n\right)}{\left(\frac{2}{T} - c_n\right)}\right] \left[z - \frac{\left(\frac{2}{T} - c_n^*\right)}{\left(\frac{2}{T} - c_n^*\right)}\right]}
 \end{aligned}$$

$$\triangleq K \frac{(z-1)^n (z+1)^n}{(z-p_1)(z-p_1^*) \dots (z-p_n)(z-p_n^*)}$$

(c.f. Section 2.4).

C.4. Derivation of Eq. (36)

From (A.6),

$$G_{BS}(s) = \frac{(s+j\omega_{a0})(s-j\omega_{a0})}{(s-c_1)(s-c_1^*)} \dots \frac{(s+j\omega_{a0})(s-j\omega_{a0})}{(s-c_n)(s-c_n^*)}$$

Using the substitution in (36), we obtain

$$\begin{aligned} G_{BS}(z) &= G_{BS}(s) \Big|_{s = \frac{2}{T} \frac{z-1}{z+1}} \\ &= \frac{(\frac{2}{T} \frac{z-1}{z+1} + j\omega_{a0})(\frac{2}{T} \frac{z-1}{z+1} - j\omega_{a0})}{(\frac{2}{T} \frac{z-1}{z+1} - c_1)(\frac{2}{T} \frac{z-1}{z+1} - c_1^*)} \dots \\ &= \frac{(\frac{2}{T} + j\omega_{a0})(\frac{2}{T} - j\omega_{a0}) \left[z - \frac{\frac{2}{T} - j\omega_{a0}}{\frac{2}{T} + j\omega_{a0}} \right] \left[z - \frac{\frac{2}{T} + j\omega_{a0}}{\frac{2}{T} - j\omega_{a0}} \right]}{(\frac{2}{T} - c_1)(\frac{2}{T} - c_1^*) \left[z - \frac{\frac{2}{T} + c_1}{\frac{2}{T} - c_1} \right] \left[z - \frac{\frac{2}{T} + c_1^*}{\frac{2}{T} - c_1^*} \right]} \dots \\ &\triangleq K \frac{[z-z_0]^n [z-z_0^{-1}]^n}{\prod_{i=1}^n (z-p_i)(z-p_i^*)} \end{aligned}$$

(c.f. Section 2.5).

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