

A COMPUTER-AIDED METHOD FOR SLEEP SIGNAL CLASSIFICATION

by

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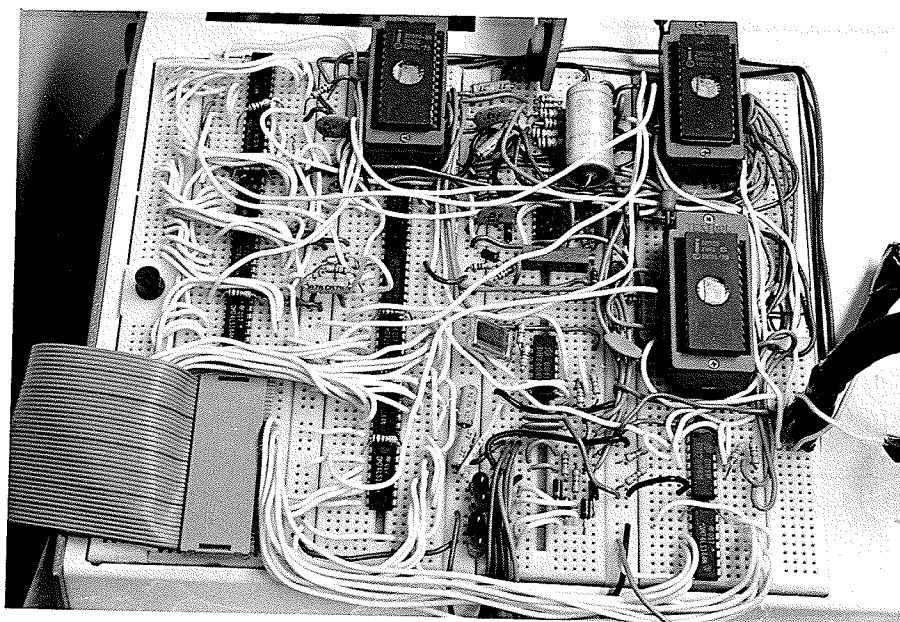
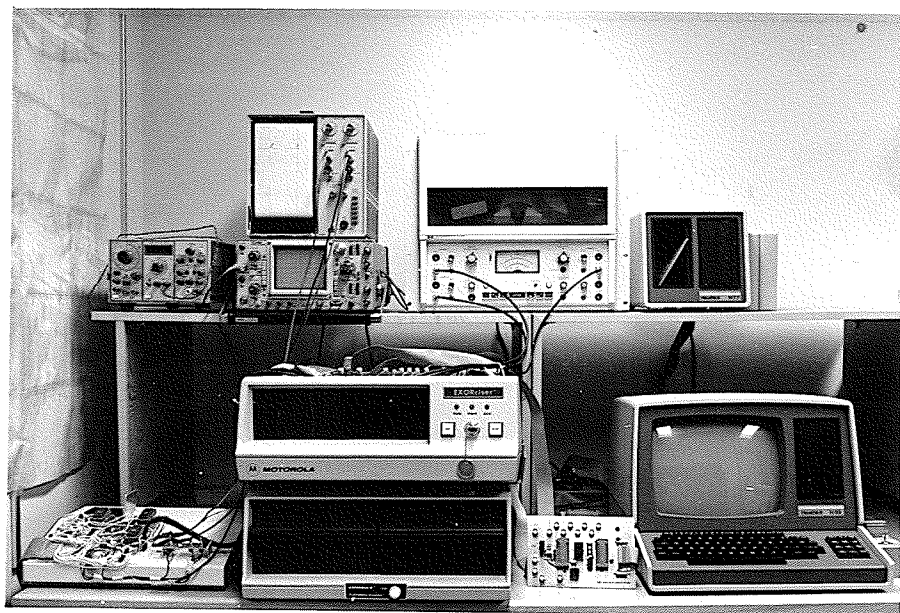
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MASTER OF SCIENCE

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


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ABSTRACT

This study considered the development of an aid to sleep classification system, and discussed a method which used the clinician as feedback to the classification process. This scheme allowed the the clinician to effect the heuristic process of sleep classification, while allowing the computer to reduce the effort expended in the classification of like segments.

The use of inexpensive microprocessing hardware for sleep signal spectral estimation was considered, and a scheme using digital filters implemented with Intel 2920s was developed. The estimation subsystem was interfaced with a Motorola EXORcisor microcomputer for real-time data collection, and the collected estimates transferred to an Amdahl main-frame computer for post-analysis.

Preliminary studies showed a gross correlation between the extracted spectral estimates and the manually classified sleep record. The spectral estimates, as sampled once a second, were found to be non-stationary within manually classified sleep stage segments. The use of spectral averaging was considered, and found to simplify the correlation between the spectral estimates and the manual classification for normal sleep. For abnormal sleep, however, it was found that such averaging could blur the correlation.

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Chapter I

INTRODUCTION

1.1 BACKGROUND

The analysis and evaluation of human sleep has become a valuable aid in the diagnosis and treatment of sleep related disorders, and is widely used today. The United States Food and Drug Administration, for example, requires that sleep studies be used in the evaluation of all new hypnotics and sedatives [1]. The evaluation of sleep is also very useful in the investigation and assessment of sleep related disorders. Insomnia, narcolepsy, chronic hypersomnia, obstructive sleep apnea, sudden infant death syndrome, and many other disorders have been studied with sleep evaluation [1,2,3]. It has been speculated that with increasing understanding and related technical advances, the evaluation of sleep can become a standard clinical procedure [1].

Although sleep has been investigated scientifically for more than 100 years, most activity has occurred in the past 25 years. The evaluation of sleep state is based primarily on the electroencephalogram (EEG), the electrooculogram (EOG), and the electromyogram (EMG). There have been many studies which have examined these signals as they occur during sleep, and these studies have led to a set of commonly accepted standards and rules for classifying sleep [1,4]. These rules define the sleep stages numbered 1 through 4, as well as the rapid eye movement (REM) and awake sleep stages. Traditionally, the sleep record is divided

into fixed (time) length segments or epochs, and the record is staged or scored on an epoch-by-epoch basis. This scoring is made primarily on the basis of the spectral distribution of the waveforms within each epoch.

In one of the largest sleep studies conducted, the rules for scoring sleep were exercised through the classification of over 6000 recordings of sleep periods for several hundreds of patients [1]. These and other studies have revealed that the signals observed during sleep, while characteristic to the rules of scoring, are quite variable with each patient [5]. Even the signals observed within one sleep session can vary significantly. The variability of these signals can arise from a variety of physiological differences as a function of patient age and state, as well as monitoring problems in electrode contact and noise.

The value of sleep evaluation as a clinical technique is offset by the labour-intensity and tediousness of the manual scoring of sleep signals recorded on paper. This labour-intensity represents a significant barrier in the investigation of sleep related disorders. The classification of a severely disturbed (8 hour) sleep record can require up to one man-week of effort. There have been many attempts to automate the scoring of sleep signal records [6,7,8,9,10,11,12,13,14]. Such automation is desirable to reduce the labour-intensity of the activity as well as to reduce the variability of interpretation from clinician to clinician [15]. The automation of the sleep signal classification process is difficult, however, and past attempts have achieved only limited success and marginal acceptance [15].

Most attempts at automated sleep signal classification have focused on the use of large main-frame computers and analog signal processing

equipment in the activity of classifying normal sleep. A number of factors have contributed to the limited success of these attempts. Classification of sleep is based primarily on the spectral distribution observed in the sleep signals [4]. This implies that the frequency spectrum of these signals must be estimated. In the past, such estimation was feasible only with delicate and expensive analog hardware or powerful main-frame computers. The large costs associated with such schemes have been a significant factor limiting their adoption. These costs are dropping, however, as recent developments in microelectronics have made inexpensive digital implementations of spectral estimation hardware possible.

A more serious problem has been that of developing a system which is capable of accommodating the high variability which is present in signals observed during sleep. This problem is similar to that of speech recognition. In both problems, it is difficult to construct a system which is able to classify observed signals at a given moment of time for a broad class of subjects. The clinician's evaluation of a sleep record is based on a wide range of experience and understanding of signal considered. Often the clinician must scan back and forth through a sleep record in order to locate representative segments of each sleep stage. These representative segments are used as reference points in the relational classification of the record. When a particular segment of a patient's record is classified, it is desired that 'similar' segments be classified in the same way throughout the record. The term 'similar' implies similarity in the most general sense; that is, a generalized metric considering the syntactic or context relations of the segment. It

has been observed that the clinician frequently deviates from the rigidly defined scoring rules for reasonable but poorly defined reasons [15]. The heuristic methods used by clinicians are difficult to describe. Most attempts at automatic sleep staging, however, have pursued a search for a simple, causal model for sleep classification. In other words, they have attempted to classify the sleep record with no perception of the record as a whole, or input from the clinician [7]. This implies that these studies have attempted to have the computer replicate the activity of the clinician with less information than the clinician requires. It can be argued that with the current state of technology and understanding of sleep, it is not possible to replicate the entire activity of the clinician [7].

Another significant limitation of these studies is that they have focused on the classification of normal sleep. Normal, healthy sleep has the characteristic that sleep stage transitions typically occur over extended time intervals. As a result, most studies have based their classification scheme on a fixed-length epoch rule. The entire night's record is divided into fixed length periods with only one decision or classification made over each interval. Epoch lengths of 30 seconds to 2 minutes have commonly been used. It has been found that patients with sleep disorders can demonstrate cyclic disruptions or periodicities in their sleep patterns. These changes can occur as frequently as every 15 seconds. For such patients, the application of a fixed-length epoch rule may prohibit the identification of brief sleep segments. An investigator may wish to relate sleep to some other physiological variable (respiration, for example). If the resolution of the sleep classification is

much less than the time constant associated with the other physiological variable, the correlation of the two functions may be lost. While past studies have achieved controlled successes on fixed healthy patient groups, the direct applicability of the work to a broad class of patients is questionable.

1.2 OBJECTIVES OF STUDY

The approach to sleep signal classification considered in this study is the use of the computer to assist the clinician in the process of classifying sleep. As in past studies, the computer will be used to extract features from the signals which are presently used to stage sleep. Then, however, the clinician will be used to identify exemplary segments to the computer for like classification. In this way, the experience of the clinician is used to effect the heuristic process of sleep signal classification, and the strength of the computer is used to reduce the effort expended in the classification of repeated segments.

The method under study represents a significant departure from past attempts in the use of computers for sleep classification in that the experience of the clinician is used as feedback to the system. Since it is desired that like record segments be classified in a like manner, and since sleep patterns tend to repeat themselves in cyclic patterns, the use of such a computer-aided system should provide a reliable method which reduces the labour-intensity of classifying normal and abnormal sleep records. The variability of the sleep record will be accommodated, as only like segments will be classified alike. If there is a high degree of variability within a given patient's record, the clinician will

have to identify more exemplary segments in order to classify the record. If a patient presents a record which has no repetitious segments and high variability, the system will effectively reduce to the manual classification method commonly used now. Areas of the record which the system has been unable to classify with the provided definitions can be related to the clinician for further clarification.

Beyond the specific application of sleep signal classification, this method represents a powerful approach for a wide variety of problems involving extended duration signal characterization and feature extraction.

The objectives of this study are to:

1. Study the digital signal processing required to effect sleep signal spectral estimation;
2. Evaluate the use of recent low-cost microprocessor hardware for the task of such estimation;
3. Develop an interface between such estimation hardware and an existing microcomputer;
4. Study the systematic procedures for evaluating an aid to sleep classification system; and
5. Conduct a preliminary evaluation of the correlation between the characterizations of the sleep signals and the manually classified sleep record.

Chapter II

SLEEP CLASSIFICATION METHODS

2.1 CHARACTERISTIC FEATURES AND RULES

Sleep signal classification rules commonly used today are based primarily on the scoring criteria established by Dement and Kleitman in 1957 [16]. These rules were reaffirmed by the U.S. Department of Health in the Rechtschaffen and Kales (1968) standardization manual for sleep scoring [4]. These studies reflect the firmly established fact that sleep is not a homogeneous state, and that sleep stages follow fairly orderly cyclic patterns [4]. Sleep can be subdivided into two broad classes of REM (rapid eye movement) and non-REM sleep. Non-REM sleep is further subdivided into sleep stages numbered 1 through 4, or as awake sleep. These definitions are based on characteristics observed in the electroencephalogram (EEG), the electromyogram (EMG) and the electrooculogram (EOG). The EEG signal electrodes are commonly applied at positions C4/A1 and or C3/A2 according to the international 10-20 placement system [17]. It is often desirable to record 2 channels of EEG, having one as a backup in case a pair of electrodes cease to function or exhibit excessive amounts of artifact. The EOG is derived from electrodes placed around the eyes, and the EMG is often taken from an electrode placed under the chin. These signals are generally recorded with a polygraphic recorder with a minimum paper speed of 10 mm/second, and a minimum sensitivity of approximately 5 microvolts per mm. The classifi-

cation of sleep is based primarily on the spectral distribution of the EEG within established bands as a function of time [4]. The spectral bands which are considered most important are the low (0.25 to 2.0 Hz) and mixed (2.0 to 7.0 Hz) frequency bands. The appearance of sleep spindles (rhythmic bursts of at least 0.5 second duration in the 12 to 14 Hz range) and K complexes (well delineated negative sharp waves which are immediately followed by a positive component [4]) are also considered significant. The EOG is commonly considered only within the band of 2.0 to 10.0 Hz, and the EMG is considered primarily from the perspective of average signal power.

Just after a patient has fallen asleep, the EEG activity is quite similar to that present when the patient is awake. This period represents the highest neurological state during sleep, and is referred to as awake sleep. The awake sleep stage has the EEG characteristic of predominately high frequency activity with some low amplitude, mixed frequency activity. Most of a night is spent in non-REM sleep, with some intermixed periods of REM sleep. During REM sleep, the EEG activity reflects a fairly 'light' level of sleep (stage 1), yet there is only a low amplitude EMG in conjunction with episodic rapid eye movements (REMs). REM appears to correspond with a 'deep' level of sleep in that there is little EMG activity, while the EEG activity corresponds with that of light sleep. As a result, REM sleep is sometimes referred to as paradoxical sleep, and is commonly thought to be associated with dreaming. Sleep stages 1 and 2 can be grouped as mixed frequency sleep, and represent 'light' sleep. Sleep stages 3 and 4 can be grouped as slow wave sleep, and represent 'deep' sleep. Stage 1 is defined as an interval of rela-

tively low amplitude, mixed frequency EEG with absolute absence of REMs, K complexes, or spindles. Stage 2 is defined as relatively low amplitude, mixed frequency EEG background with the occurrence of sleep spindles and/or K complexes. If more than 3 minutes of stage 2 sleep pass without a sleep spindle or K complex, the classification defaults back to stage 1. Stage 3 is defined by an EEG segment in which at least 20% but not more than 50% of the epoch consists of predominately low frequency activity. Stage 4 is defined by an EEG segment in which more than 50% of the epoch consists of primarily slow frequency activity. It is often difficult to differentiate between stages 3 and 4, and they are, therefore, sometimes classified together as slow wave sleep.

Normal sleep pattern distributions vary dependent on factors such as as patient age and sex [1]. Sleep patterns can also be altered by a variety of clinical disorders. An example of a disorder which can alter sleep is sleep apnea syndrome [2]. Sleep apnea is a disruption of respiration where breathing is diminished or stops for periods exceeding 10 seconds. During such events, the oxygen content of the patient's blood may drop. Events of this nature are typically terminated with an arousal, although the patient is generally not aware of the event. Frequent disruptions of sleep may lead to variety of symptoms or disorders, and can have a significant impact on the patient [2].

2.2 MANUAL CLASSIFICATION METHODS

The most commonly accepted method for sleep signal classification is a manual review of the polygraphic recording. Traditionally, the entire night's record is divided into fixed-length sections or epochs. Epoch lengths of 30 seconds to 2 minutes have commonly been used. Classification is based on the visually observed waveforms averaged over each epoch interval. A single classification is made for each epoch, and the record is classified on an epoch-by-epoch basis. The clinician will often scan through the entire record in order to 'calibrate' for the scoring, and must often refer to other portions of the record in order to classify a particular epoch. For the classification of abnormal sleep, a variable length epoch method has been used [18]. With this scheme, sleep stage transitions are identified wherever they may occur (with a minimum epoch length of 15 seconds). This method allows a more accurate tracking of disturbed sleep patterns. The classification data is then commonly entered into a computer for further analysis. The classification data has, for example, been correlated with respiration variables concurrently recorded during sleep [19].

2.3 AUTOMATED CLASSIFICATION ATTEMPTS

There have been many attempts to automate the classification of sleep signal records. These attempts have used a variety of signal processing techniques, and have considered many different decision algorithm methods. In general, these attempts have focused on trying to achieve total automation of the sleep classification process. In a review of attempts to automate sleep classification [15], it was noted that most studies

that have achieved some satisfactory results have had much in common. These systems have, in general, used REM detectors to discriminate stage 1 from REM, and have tried to establish some algorithm to mimic the heuristic decision making of a clinician on the basis of some wave features extracted from the EEG. The EMG, although valuable according to the Rechtschaffen standards, has often not been considered as it has been found to be unreliable [15].

Direct amplitude analysis, spectral analysis, Walsh expansions, and period analysis have all been considered in the problem of sleep classification [20,21,22]. Although different studies have used widely varied signal processing techniques and implementations, the more advanced attempts have achieved approximately the same level of controlled success. Frequently used methods for the wave feature extraction have been band-pass filtering, fast Fourier transforms (FFTs), matched filtering or correlation detection, zero crossing and pitch period detection, or the use of hybrid wave-form detection circuitry. The most frequently used technique for the wave feature extraction has been bandpass filtering. The filters have, in the past, commonly been implemented with analog circuitry. The low frequency nature of the passbands has made such equipment difficult and costly to implement. Many other studies have recorded the entire sleep record and have conducted post-analysis FFT studies. There has been much study of the detection of sleep spindles, K complexes, and rapid eye movements (REMs) [23,24,25,26,27,28,29,30,31]. K-complexes are frequently not considered as they are poorly defined and difficult to detect in both manual and computer-aided analysis. REMs are very valuable in distinguishing be-

tween stage 1 and REM, and sleep spindles are also generally considered. Overall, the primary factor considered in sleep classification is the amplitude of the EEG within spectral bands as distributed over time.

Independent of the wave features extracted, the first study to accurately classify all stages of sleep was a scheme which incorporated pattern recognition [15]. The early work with pattern recognition was criticized because it could not accommodate a patient group with a large variance in age without changing the recognition scheme. Also, these systems often consumed very large amounts of expensive computer time to effect the wave feature extraction and pattern recognition. Pattern recognition is a field which has received much study of late, however, and holds promise in its application to a number of areas including sleep classification [7,6,32]. Many of the other studies which achieved some success used decision tree algorithms to try to directly mimic the activity of the clinician. These studies were frustrated by the very heuristic nature of manual sleep classification, and as a result achieved success only within confined patient groups.

Chapter III

PROPOSED METHOD

The general method considered in this study is one of having the computer aid the clinician in the classification of sleep. The computer is used to extract the wave features which are considered significant in the classification of sleep, and the clinician identifies to the system exemplary segments of the sleep stages. In this way, the experience of the clinician is used to effect the heuristic process of sleep classification, and the computer is used to reduce the effort expended in the classification of like segments. When a particular segment of a sleep record is classified, it is desired that like segments be classified in a like manner. Since sleep patterns tend to repeat themselves during the course of a night, a system which could identify repeated segments could reduce the labour-intensity of the activity. This is especially true for the classification of disturbed sleep in that, for patients with such disorders, the frequency of sleep stage transitions is greatly increased. In a more general sense, this method represents a powerful approach for a wide variety of problems involving extended duration signal characterization and classification.

The conjecture of this approach is supported by much of the work done on automated sleep classification. Gath and Bar-on [7] studied a system which extracted linear predictive coding (LPC) coefficients for the EEG, EOG, and EMG during sleep in order to characterize the entire nights ac-

tivity. The spectrum of the signals was then estimated on the basis of the LPC coefficients, and a scheme using fuzzy set theory was used to classify the sleep (fuzzy set theory is closely related to pattern recognition). This work reinforced the conjecture that the average spectrum presented in the EEG is 'somewhat' stationary during sleep. The term 'somewhat' in this case implies that the average spectrum is constant for some time, although varying over longer time intervals. In fact, this work suggested that the transitions in the average signal spectrums were quite distinct, and that an 8-hour recording could be reduced to about 3500 record segments with somewhat constant average spectral characteristics. Bourne [6] has extensively studied the use of syntactic pattern recognition for sleep classification, and has had promising results.

The primary reasons for considering a feature extraction/pattern recognition scheme are because the characteristics observed during sleep vary greatly between patients, and because the classification process effected by the clinician is so heuristic. This is not to criticize clinicians for their inability to describe completely the process which they effect. An analogy might be that an individual, on hearing two distinct pieces of music, is clearly able to distinguish between them although generally unable to say exactly why. What is implied is that the human brain is an extremely capable processing element which is able to conduct very heuristic decision making. Given the heuristic nature of the sleep classification process, and the fact that repeated attempts to establish decision-tree like structures to model the process have achieved only marginal success, a generalized pattern recognition scheme

which bases the classification on identified exemplary segments seems like an appropriate avenue for consideration. The consideration of highly variable, disturbed sleep records further intensifies the need to move away from a fixed-model approach.

Some of the studies which have considered pattern recognition have been found to be very intensive in their computer usage. These costs have limited their use. The ever increasing complexity and dropping costs of microprocessing hardware provide strong motivation for their use in problems like sleep classification. Recent developments in microelectronics have led to microprocessors well suited for use in signal processing. The Intel 2920 is an example of such a microprocessor [33]. The use of such microprocessing hardware for the task of sleep classification may lead to the development of an inexpensive labour-saving aid. Therefore, this study has considered the use of inexpensive microprocessor hardware for sleep signal classification.

Chapter IV

SLEEP SIGNAL PROCESSING

4.1 INTRODUCTION TO DIGITAL SIGNAL PROCESSING

Analog signals are those which are continuous as a function of time. Digital signal processing is the general subject associated with sampled signals which are represented numerically. Sampling a signal and processing it numerically has a number of advantages over processing a signal with analog components. Numerical processing is frequently accomplished with a digital computer. This fact implies that there is a high degree of flexibility in the processing which can be accomplished by adjusting software algorithms. Achieving such flexibility with analog hardware is, in general, not economically feasible. When an analog circuit is specified, the components with which it is constructed are only approximations of the values desired. This implies that precise analog circuits require fine tuning after the components particular to the circuit have been installed. Also, after the components have been installed and fine tuned, the circuit function can drift as the analog components age and change in characteristics. This drift may require that the circuit be periodically fine tuned to reestablish its function. Independent of the approximation and drift problems, it may be difficult to fabricate the components required in an analog circuit. This problem is often true when working with low frequency signals. Also, whenever an analog signal is processed, there is almost always some noise added to

it. Systematic and expensive design and fabrication procedures can minimize (although not eliminate) these problems.

Digital signal processing does not solve the problems identified with analog processing, but rather replaces them with problems that can sometimes be more easily addressed. In general, digital components which are stable and drift free can be fabricated quite accurately and repeatably. Also, functions which may be difficult to implement with analog components are often straightforward with digital signal processing techniques. Once a signal is sampled, there is no additive electrical noise since the processing occurs through numerical operations. Like most things in life, however, there are tradeoffs to counter the benefits realized in using digital signal processing techniques. Today, most digital signal processing hardware implementations are more costly than their analog counterparts (unless very delicate, noise immune, or low frequency analog circuits are required). In place of additive electrical noise in processing a signal, finite register quantization error, round off and truncation errors, and register overflow errors introduce noise into the signal. Also, the cost of the digital hardware system increases sharply if real-time high frequency signal processing is required. Therefore, digital signal processing is not an automatic choice over its analog counterpart.

4.2 CONSIDERATIONS IN SAMPLING A SIGNAL

Sampling a signal, or establishing a digital representation for the value of a signal at a specific moment in time, is generally accomplished with an analog to digital (A/D) converter. Since A/D converters require a finite time to perform such a conversion, an analog sample and hold circuit is often used as a buffer between the input signal and the A/D converter. This circuit acquires the value of the continuous signal at the moment of sampling, and holds the value constant while the A/D conversion is completed. The digital signal processing algorithm which is applied to such a sampled signal depends on what processing function is desired. Whatever algorithm, the function of the process is implemented through a sequence of numerical processes, in general operating on the sampled or quantized input. Since the process is effected through numerical operations, a digital computer is ideally suited for the implementation of such algorithms. The processing of such an input signal may occur while the signal is occurring (in real-time), or the signal samples may be stored for later processing.

The type of result which is obtained from a digital signal processing algorithm depends on the algorithm being performed. The results may be processed further by other algorithms, or the result of the algorithm may be a stream of digital values which correspond to a transform of the analog input signal. It may, therefore, be desired to reconstruct an analog output signal from such a stream. This function is accomplished with a digital to analog (D/A) converter. A D/A converter produces an analog output which is proportional, at a given moment, to the digital value which is supplied to it.

The sampling of a continuous signal affects the frequency domain characteristic or spectrum of that signal, depending on the relationship of the sampling rate to the highest spectral component of the signal. Intuitively, it seems reasonable that if a signal is varying 10 times a second, sampling the signal once a second will not 'capture' the information in the signal. The Nyquist relation states that, for an infinite duration signal, sampling at twice the frequency of the highest spectral component of the input signal will result in a complete capture of the information contained in the signal. Sampling at rates below the Nyquist rate results in aliasing distortion or noise. This relationship is best explained by a diagram. In Fig. 4.1, it can be seen that the spectrum of the sampled signal is that of its analog counterpart, except that the spectrum repeats itself or is folded by the sampling frequency. If a sampling frequency of less than two times the highest spectral component of the input signal is used, aliasing noise is added into the sampled spectrum. In practice, sampling rates well in excess of twice the highest component are used. Bandlimiting lowpass or anti-aliasing filters are often used to bandlimit the input signal and ensure a minimum of aliasing noise.

Since the continuous signal is represented by digital values of finite length, quantization noise is also introduced. If an N bit representation is used, 2^N values can be represented. If a signal variation smaller than $1/(2^N)$ of the input signal range occurs, the variation will go 'unnoted'. This problem manifests itself as additive quantization noise which is a function of the number of bits of resolution used.

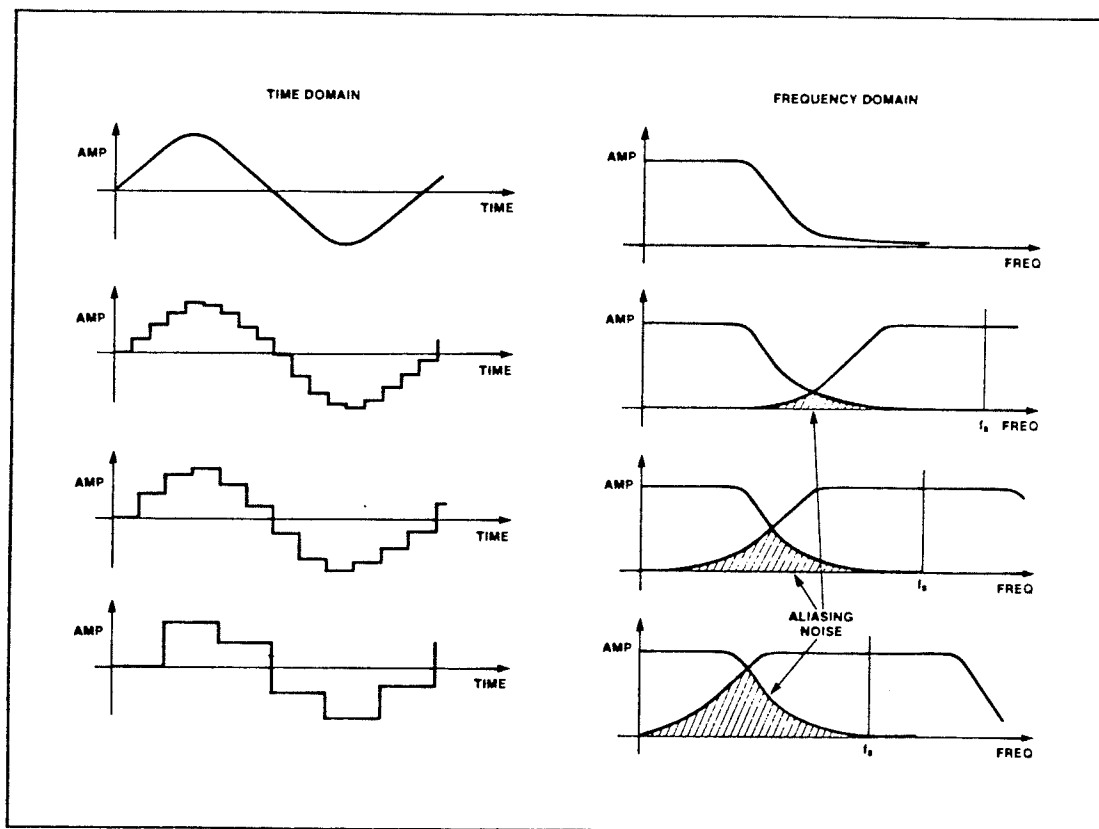


Figure 4.1: Spectrum aliasing with decreasing sampling frequency

Taken from Intel Corp., 2920 Analog Signal Processor Design Handbook. Santa Clara, 1980.

If an analog output is required from a digital system, this is accomplished with a D/A converter. The output of such a system remains constant until a new output is available. At that time, the output jumps to the new output value. These jumps introduce high frequency components into the output. A lowpass reconstruction filter is often used to smooth the output waveform.

4.3 SPECTRAL AMPLITUDE ESTIMATION FOR EEG AND EOG

The first step in any system which is to aid in the sleep signal classification process is the consideration of spectral amplitude estimation. The most frequently used method for such estimation has been analog bandpass filtering. The low frequency nature of these signals, however, makes analog solutions difficult and costly to implement. As a result, various digital signal processing techniques have been attempted with varying degrees of success. Pitch period detection algorithms have been used, and are based on a time domain analysis of a waveform. The algorithm detects the rise or fall of a signal, thereby determining when peaks and troughs have occurred. With the time between peaks and troughs and the difference between the peak and the trough, a crude estimate of the amplitude spectrum is made. These algorithms are desirable because of their simplicity, but have had only limited success when applied to the problem of EEG amplitude spectrum estimation. Fast Fourier transforms and Walsh transforms have been applied to the problem, and both have achieved good success. Between the two methods, FFTs are more commonly accepted and used. Linear predictive coding has been used as a method of characterizing the EEG and EOG in real-time, with Fourier

transform techniques applied in post-analysis to evaluate the spectrum. Correlation techniques for spectral estimation of sleep signals have not commonly been used, although recent publications on spectral analysis have advocated their use [34].

The decision on which method to use in an implementation for the spectral amplitude estimation is influenced by several factors. A significant factor steering any such practical decision is the cost of the solution. Many of the attempts at this problem have been based in large and well funded laboratories which have been able to afford main-frame computers dedicated in real-time to the problem. Less expensive but still costly solutions have been to store all the samples from an entire night's sleep and analyze the signals after the fact. This alternative is costly in that a large capacity of bulk storage is required for the samples (at least 1.5 Megabytes formatted per EEG track).

Microprocessor based solutions are comparatively inexpensive. Most inexpensive microprocessors, however, have instruction sets and word sizes which are not well suited for digital signal processing applications. As a result, the use of microprocessors for real-time digital signal processing can be difficult. More advanced configurations of general purpose microprocessors can be used, but recent advances have also been made in the development of microprocessors which are oriented towards digital signal processing. The Intel 2920 is such a device, as it contains analog multiplexers, sample and hold circuitry, an A/D converter, random access memory (RAM), a programmable permanent memory (EPROM), a D/A converter, and output demultiplexer and hold circuitry on one integrated circuit (IC). The 2920 is able to implement a shift and

add/subtract in one instruction, and is able to conduct parallel digital and analog operations. All internal registers are 24 bits wide, with a 9 bit bipolar A/D and D/A converters. This single chip signal processor is quite suitable for real-time signal processing. Real-time signal processing reduces the need for a bulk storage device to save the samples, thereby significantly reducing the cost of the system.

Given the attempt to try to reduce the cost of the system through real-time signal processing with the Intel 2920, it is necessary to review the processing requirements of the various algorithmic alternatives. The fast Fourier transform is an algorithm which performs a time domain to frequency domain transformation. Given a set of time domain samples, this algorithm provides a transformation to a set of values which correspond to estimates of the frequency domain characteristics presented in the time domain samples. The results of this algorithm are, in general, complex values which estimate the magnitude and phase of the spectral components. The larger the set of samples taken in such a calculation, the better the resolution or spacing between the values calculated in the frequency domain. The FFT, however, is subject to a phenomenon known as windowing. Briefly, there is distortion in the calculated spectrum which results from the fact that the input signal has not been considered for an infinite time. Any real FFT algorithm is of finite size, and therefore windowing distortion is always present in the calculated spectrum. Advanced techniques can be applied to reduce this problem, although it can not be totally removed. The FFT is also quite processing intensive. In order to perform the FFT, a large number of complex arithmetic calculations, and enough memory to store at least

the samples for the FFT and the results are required. Although the 2920 is limited in the amount of available memory, and does not implement complex arithmetic very easily, it is well suited for digital filtering. Digital filtering is a technique which facilitates a digital approximation of an analog filter. Primarily for this reason, together with the experience in EEG analysis based on spectral estimation within bands, digital filtering was chosen as the method for the spectral estimation.

4.4 SPECTRAL ESTIMATION USING DIGITAL FILTERS

The scheme used to estimate the amplitude spectrum of the EEG using digital filters can be described as follows. A set of bandpass filters is used to separate out the spectral bands of interest from the incoming EEG and EOG signals. The bands which have been found to be of significance are: 0.25 to 2.0 Hz, 2.0 to 7.0 Hz, and 12 to 14 Hz for the EEG, and 2.0 to 10.0 Hz in the EOG. In order to establish reasonably narrow passbands, fourth order Butterworth bandpass filters were chosen to pass the segments of interest. Since the passbands all exist above dc, the dc component of each of the output waveforms should equal zero. The output from each of the bandpass filters is then passed through a full wave rectifier. The output of the full wave rectifier is then passed through a low pass filter with a break frequency very near dc. The effect of the scheme is that the output of the low pass filter is a value which corresponds to the dc or average in the rectified bandpass signal. A block diagram of this process is shown in Fig. 4.2. In an attempt to minimize the ripple from the rectified bandpass in the low pass filter output, break frequencies for the low pass filters were chosen at least a factor

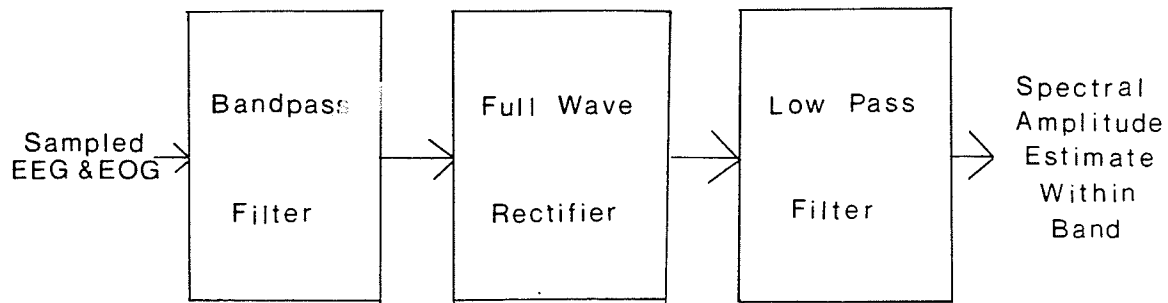


Figure 4.2: Process block diagram

of 10 less than the lower break frequency of the bandpass filter. Also, second order Butterworth lowpass filters were chosen to reduce the ripple in the output.

4.5 FILTER DESIGN CONSIDERATIONS

Infinite-Impulse Response (IIR) digital filters are those which have a response of infinite duration to an impulse input. An IIR filter has a transfer function, $H(z)$, of the form:

$$H(z) = \frac{b_0 z^N + b_1 z^{N-1} + b_2 z^{N-2} + \dots + b_N}{z^N + a_1 z^{N-1} + a_2 z^{N-2} + \dots + a_N} \quad (4.1)$$

where all the a 's are not zero. IIR filters are desirable because they provide good transfer characteristics from a comparatively simple structure. The feedback in these structures, however, introduce the possibility of the filter being unstable. The design of the digital filter is generally accomplished through a transformation of an analog filter design. The design objective is to arrive at a filter design which meets not only the specifications, but is also stable, causal, and is simple enough to be realized with available hardware.

Specification of a filter can include amplitude and phase characteristics, as well as transient performance specifications. There are basically three types of transformation methods which can be used to transform an analog filter design to a digital design. Each of these methods has specifications which are ensured in the digital design. The impulse-invariant and step-invariant methods ensure that the resulting digital filter will have the same impulse or step transient performance as their analog counterparts. Amplitude and/or phase characteristics are usually sacrificed for these results. The third common method is the bilinear transform method. This method provides a filter which approximates the amplitude characteristic of its analog counterpart quite well. For this reason, the bilinear technique was chosen for the design of the bandpass and lowpass filters.

The bilinear transform method maps the left-half of the s-plane into the interior of the unit circle of the z-plane. The $j\omega$ axis of the s-plane is mapped onto the $|z|=1$ circle of the z-plane. For a stable filter in the z-plane, all poles must reside inside the unit circle. Since all the poles of a stable analog filter reside in the left-half of the s-plane, the digital filter obtained by the bilinear transformation of a stable analog filter is always stable. This mapping is obtained by substituting $z=(1+s)/(1-s)$. Since an infinite space is mapped into a finite space, the mapping cannot be linear. The analog frequencies are therefore prewarped with the tan function $\bar{\omega} = \tan(\omega T/2)$ where T is the sampling period of the digital filter. The derivations of the digital filter designs are included in the Appendix.

The choice of sampling rate is based on many factors. If the sampling rate is chosen too low, significant aliasing noise can be introduced into the spectrum. To counter this problem, a sampling rate much higher than the highest critical frequency is desirable. The sampling rate, however, also affects the positions of the poles of the filter on the z -plane. This can be seen in Fig. 4.3. For the example of the 12-14 Hz bandpass filter (4th order Butterworth), it can be seen that as the sampling rate increases, the poles are in general shifted towards the $|z|=1$ circle. Since the $|z|=1$ circle represents the limit of stability, the sensitivity of the system is increased as the poles approach that limit. Any real digital filter can only approximate a designed system to a finite accuracy. As the poles approach the $|z|=1$ limit, the possibility of the approximation residing outside that limit increases. For each of the filters used in this study, the difficulty of realizing each pole was considered. The tradeoffs between sampling rate, pole sensitivity and scaling were considered somewhat heuristically, and the resulting sampling rates are shown in Table 4.1. Complete data on filter coefficients and scaling are included in the Appendix.

TABLE 4.1

Filter characteristics and operating rates

Filter Characteristics (Butterworth): Operating Rate

1/4	-	2.0 Hz	4th order bandpass	:	50	Hz
		.025 Hz	2nd order lowpass	:	12.5	Hz
2.0	-	7.0 Hz	4th order bandpass	:	50	Hz
		0.2 Hz	2nd order lowpass	:	50	Hz
2.0	-	10.0 Hz	4th order bandpass	:	50	Hz
12.0	-	14.0 Hz	4th order bandpass	:	200	Hz
		1.0 Hz	2nd order lowpass	:	200	Hz

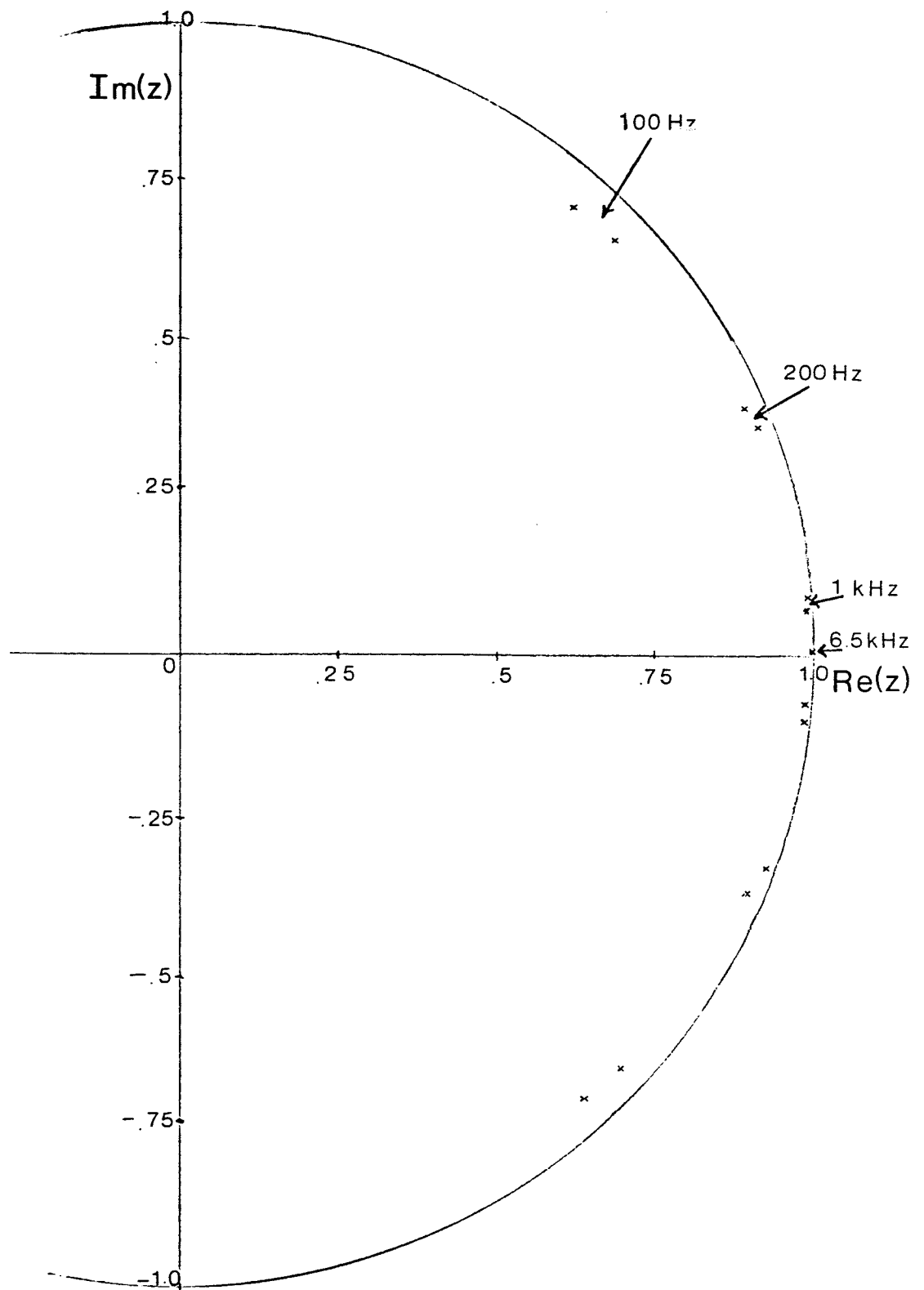


Figure 4.3: Shifting pole position with sampling rate

4.6 IMPLEMENTATION CONSIDERATIONS

The design considered in the previous section must be implemented with a structure of arithmetic operations and delays. The implementation is best described with a block diagram. These block diagrams use delays, multipliers, and adders. Transfer functions are most often realized in blocks of first or second order sections, since higher order sections are quite sensitive to parameter variations. Direct or canonical-forms are two common realization methods. The poles associated with the filters are neither close to the $|z|=0$ point or the $\text{Im}(z)=0$ line. Given this fact, and the fact that there are complex conjugate poles, Chen [35] recommends the use of parallel (second order) canonical-form realization blocks. This choice is based on a comparison of the sensitivity of the poles, the number of arithmetic operations required, and the flexibility to realize complex conjugate poles with each form. The transfer functions for the filters are:

$$\text{Bandpass: } H(z) = \frac{K(z-1)(z+1)}{z^2 - Az + B} \cdot \frac{(z-1)(z+1)}{z^2 - Cz + D} \quad (4.2)$$

$$\text{Lowpass : } H(z) = \frac{K(z^2 + 2z + 1)}{z^2 + \alpha p z + \beta t} \quad (4.3)$$

These transfer functions can be realized with the block structure seen in Fig. 4.4.

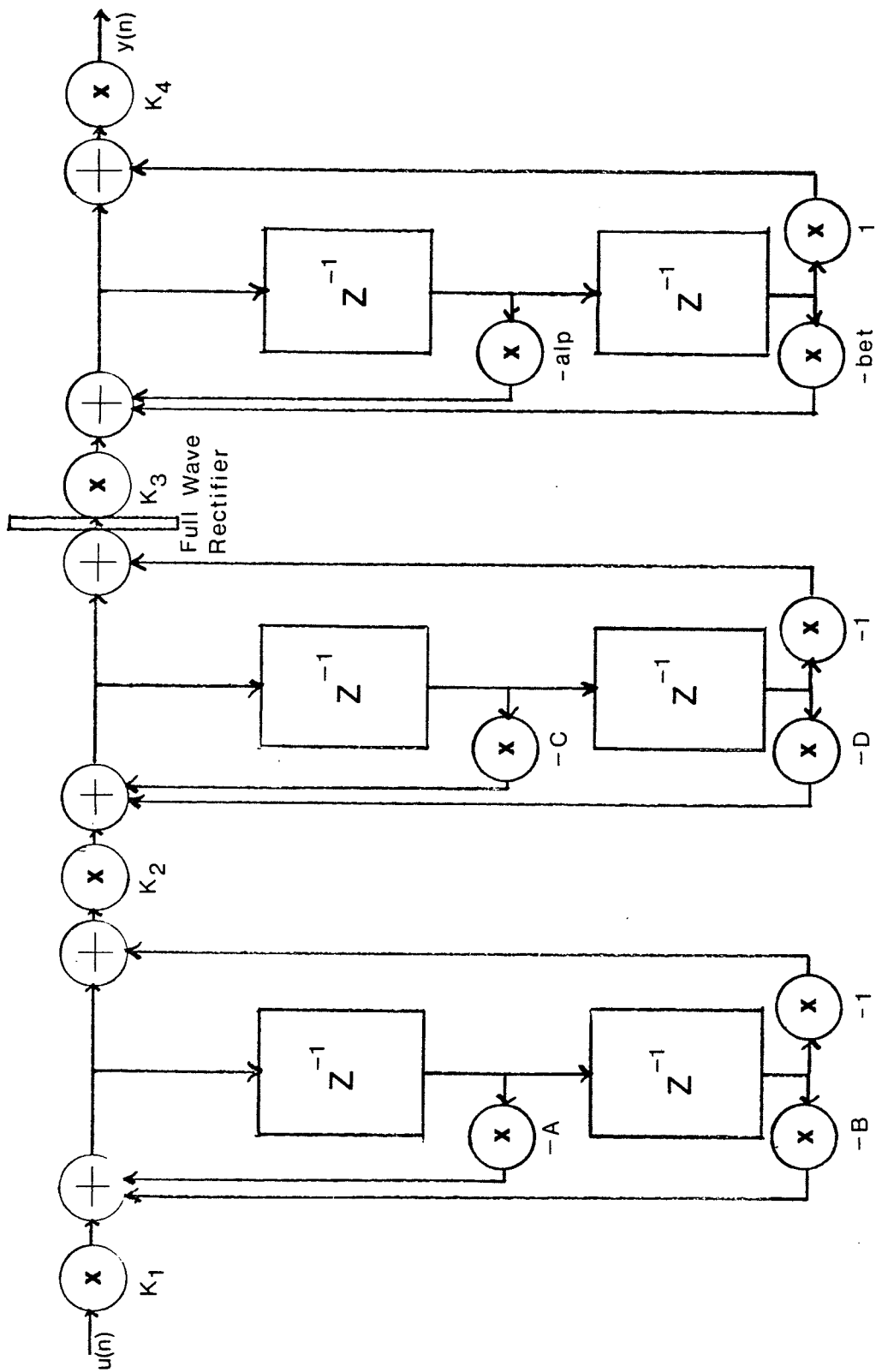


Figure 4.4: Block realization

4.7 INTRODUCTION TO THE 2920 SIGNAL PROCESSOR

The Intel 2920 signal processor is a microprocessor designed for signal processing. The device contains on a single chip: a 9 bit bipolar A/D converter with a 4 channel multiplexer and sample & hold, a D/A converter with 8 demultiplexed output channels with sample & holds, a 25 bit arithmetic logic unit, 40 - 24 bit dual port RAM locations, 192 - 24 bit program (EPROM) memory locations, a shifter unit, and an instruction set which is designed for doing digital signal processing. Each instruction has an arithmetic and an analog part. Arithmetic and analog processes are conducted in parallel. An arithmetic shift and add/subtract can be implemented with one instruction. No internal bus signals are available, and the device is simply supplied with input signals, power, and a crystal. A diagram of the architecture of the device is shown in Fig. 4.5.

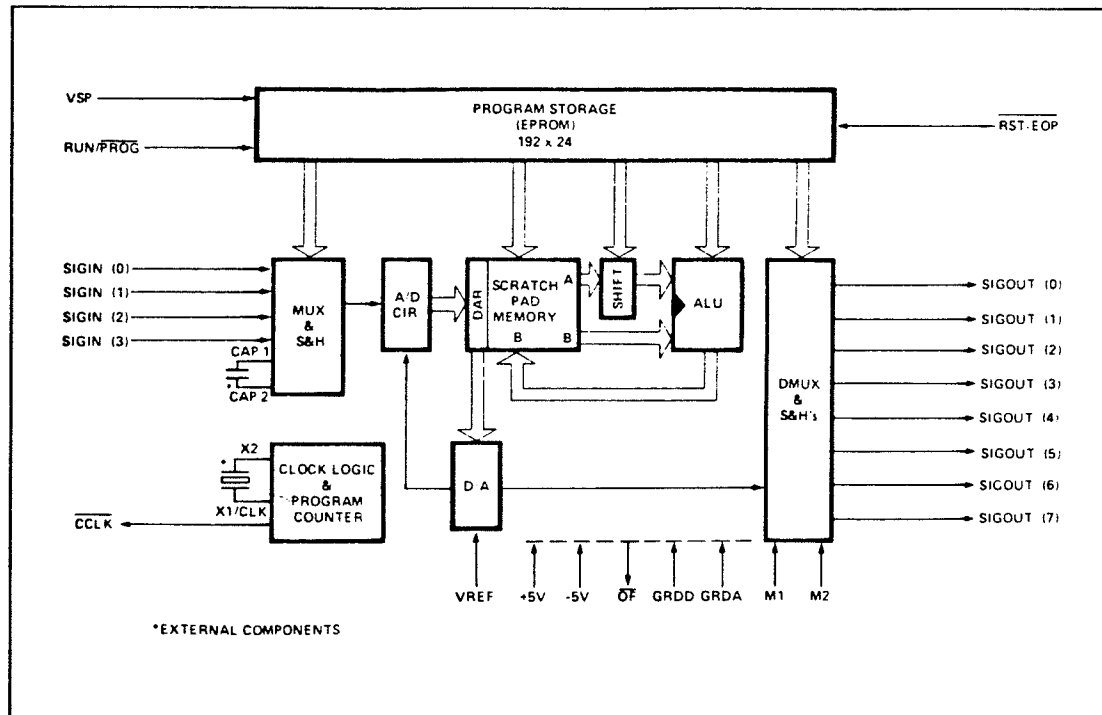


Figure 4.5: 2920 block diagram

Taken from Intel Corp., 2920 Analog Signal Processor Design Handbook. Santa Clara, 1980.

4.8 2920 PROGRAM DEVELOPMENT

The 2920 is programmed via an EPROM which resides in the device. This implies that an EPROM programming circuit must be available to program a 2920. Also, the 2920 instructions are 24 bits long and comprised of 5 operands. The bit string is also jumbled before it is programmed into the device. The effective result is that an assembler is essential to the development of 2920 code. The assembly/programming function can be purchased from Intel in two forms. The best solution is to buy an Intel development system. This option, however, costs approximately \$50,000. A second alternative is to purchase a dedicated development board from Intel for a cost of approximately \$1,400. (The University of Manitoba has purchased such a board). This board is, however, somewhat unreliable and less than optimal in terms of ease of use. Therefore, Mr. Andy Weirich and I designed and implemented a 2920 assembler and programmer for the Motorola EXORcisor laboratory computer. Programming is accomplished through a programmer board inserted in the EXORcisor system bus. The assembler was developed in Motorola MPL [36] (a PL/I-like compiler which compiles to 6800 assembly mnemonics), is 940 lines long, and operates in the EXORcisor. It allows free format entry of 2920 programs, with full error diagnostics and error recovery during assembly and programming. Programs are entered and edited with the EXORcisor editor, and are stored on floppy disk. An interface between the EXORcisor and a Heath H14 printer was also developed, to allow paper copy of disk files. A listing of the assembler source code is included in the Appendix.

4.9 FILTER PERFORMANCE CONSIDERATIONS

The coefficients for the digital filters were calculated with a FORTRAN program which is included in the Appendix. The 2920 is able to multiply with shifts and adds/subtracts, and therefore a sign digit canonic form breakdown of each of the coefficients was also evaluated by the program. The 2920 used in this implementation was supplied with a 5.0 MHz crystal. The 2920 requires 4 crystal cycles/ instruction, and therefore a maximum length program of 192 instructions results in a sampling rate of approximately 6.5 kHz (assuming only one sample is performed per program pass). At this sampling rate the poles of the digital filters are very sensitive and difficult to realize. Therefore, a submultiple sampling scheme was implemented. The submultiple sampling scheme is based on software counters and conditional delays. The calculations for all the filters are conducted at a rate of 6.5 kHz, with delays effected only when certain software conditions have been met with the counters. Through this scheme, software adjustable operating rates were established to effect each of the desired operating rates. The interested reader can refer to Intel's design handbook [33].

Anti-aliasing filters were not implemented as the EEG and EOG signals available for this processing were already bandlimited. The outputs of the digital filters were transferred to a second computer digitally, and therefore, reconstruction filters were not considered.

Overflow and scaling considerations were addressed as follows. Given a realization of a structure with a given machine, it is a nontrivial problem to establish which (bandlimited) input sequences will result in the highest probability of overflow [37,38]. In order to determine some

estimate of appropriate scaling factors, the 2920 operations were simulated on the Amdahl computer in FORTRAN. Practicing saturation arithmetic as effected by the 2920, a full scale input signal at the center frequency of each bandpass segment was passed through each of the filter setups. After operating each of the filters long enough to settle down, several cycles of the input waveform were passed to determine if overflows were occurring anywhere in the structure. If overflows did occur, the appropriate scaling factor was reduced by a power of 2 and the process repeated. In this way, a set of scaling factors was determined which provided a reasonable, though not optimal, use of the implementation hardware. A listing of the simulation software is included in the Appendix.

4.10 FILTER TESTING PROCEDURES

Each of the filter setups was tested with sinusoidal signals, with the outputs from the filters converted to analog by the 2920. The low frequency nature of the filters allowed for accurate paper plotting of the output. The performance of each of the setups was measured with standard frequency response techniques. This was accomplished by applying a sinusoidal input and measuring the amplitude of the output. The filters were found to perform very well, and the spectral amplitude characteristics for each of the filters matched that theoretically predicted almost exactly. This level of performance can be attributed, for the most part, to the large word (data) path provided by the 2920. As mentioned earlier, the 2920 provides a 9 bit A/D conversion, and uses/stores 24 bits for all calculations. Also, all calculations are conduct-

ed with 25 bits of accuracy, with the result rounded to 24 bits. The spectral amplitude characteristics for the bandpass filters is shown in Fig. 4.6.

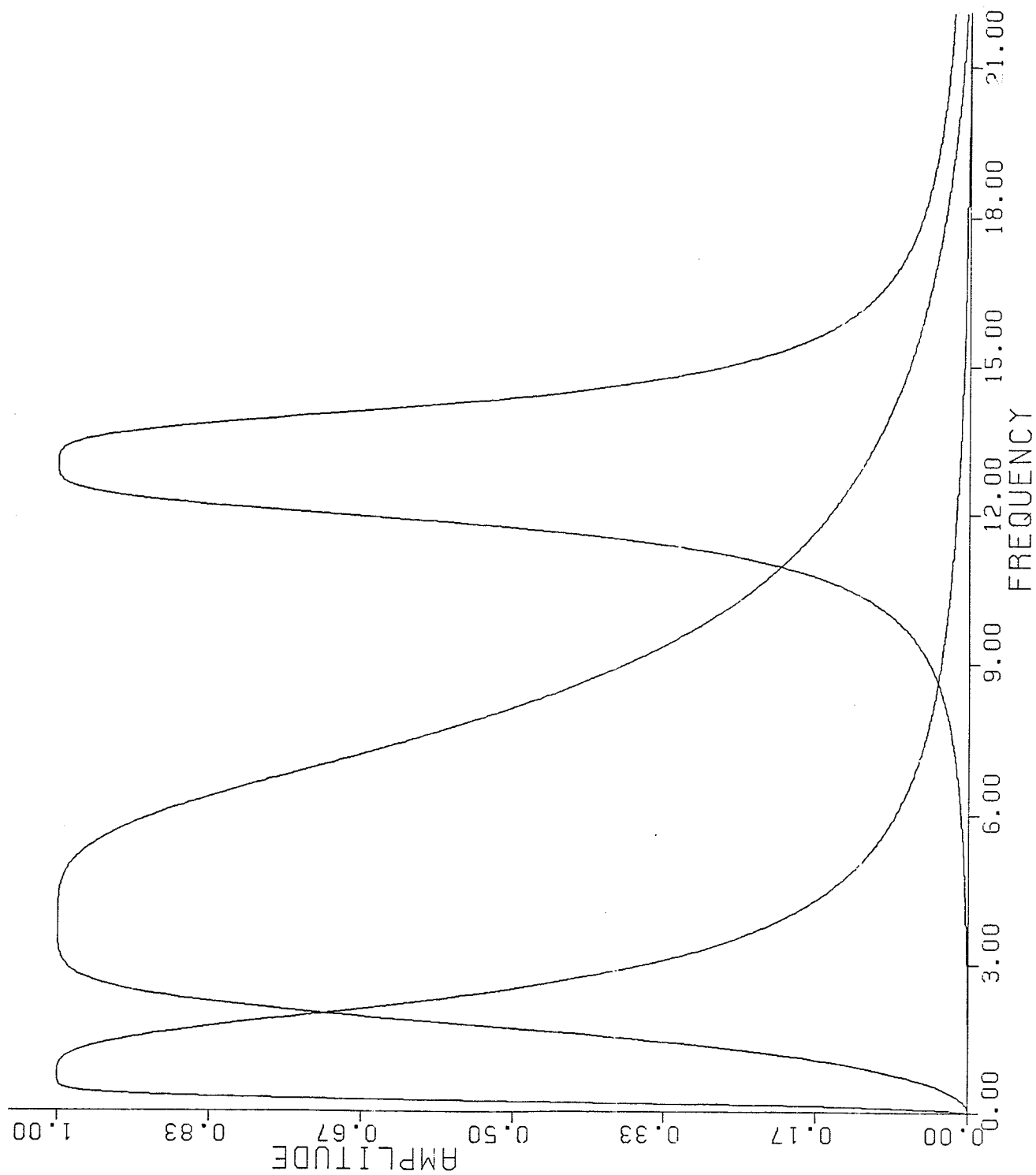


Figure 4.6: Bandpass filter amplitude characteristics

Chapter V

TEST SYSTEMS AND PROCEDURES

5.1 SYSTEM OVERVIEW

In order to implement the proposed aid for sleep classification, it is necessary to collect the spectral estimates extracted during the recording, and to store them for post-analysis. To accomplish this task, it is necessary to interface the 2920s to a second computer or intelligent storage system. With the extracted features stored, the post-processing can be conducted by the data collection processor, or the data can be transmitted to a more suitable computer.

Within the confines of this study, the objectives were to investigate the proposed method, and to study the use of inexpensive microprocessing hardware for the spectral estimation. Given those objectives and the budget constraints of the project, a system hardware scheme was devised using available equipment. For these reasons, the 2920s were interfaced to a Motorola EXORcisor laboratory microcomputer, and the collected data was transmitted to an Amdahl V7 main-frame computer. The EXORcisor is a 6800 based microcomputer with 56 Kbytes of RAM, 2 floppy disk drives, parallel and serial interface ports, and a disk operating system (Motorola EXORDisk II MDOS). The Amdahl is a large computer which supports many online users, and has a Versatec graphics plotter attached to it.

The EXORcisor does not use a dedicated floppy disk control device, and therefore the 6800 processor itself controls all disk operation ac-

tivities. This fact, combined with the time required to conduct floppy disk activities, imply that the 6800 may be occupied for periods of up to several seconds in a disk read or write. Since it was desired to sample the spectral estimates from the 2920s many times within the briefest sleep stage interval (approximately 15 seconds), a sampling rate was chosen to be 1 Hz. Given this desired sampling rate, it was not possible to conduct disk operations in real-time without missing some sample times. Therefore, it was decided to collect data only to the limit of the EXORcisor's RAM storage capacity, with storage on disk after collection. The collected data was then transmitted to the Amdahl through an RS-232 asynchronous communications link for post-analysis. A system data flow diagram of this configuration can be seen in Fig. 5.1.

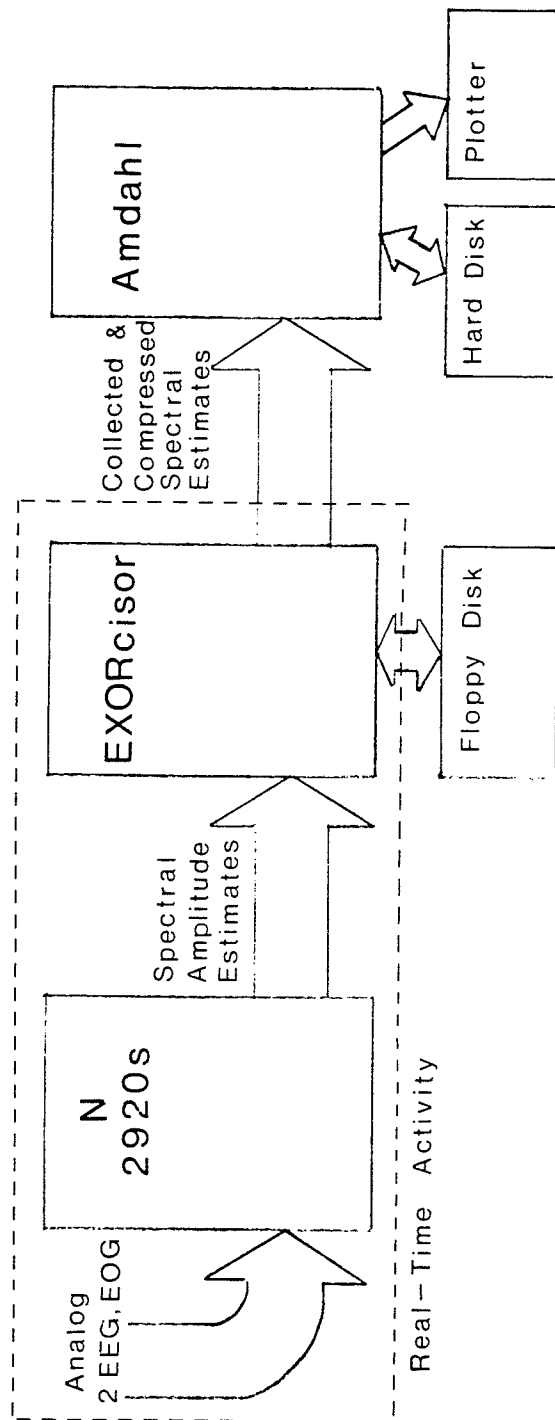


Figure 5.1: System data flow diagram

5.2 2920/EXORCISOR INTERFACE

5.2.1 Transfer Method

A 2920 is capable of producing analog or digital output from any of its 8 output lines. Since the outputs of the spectral estimation algorithms in the 2920s are numerical, it was decided to transfer the estimates digitally. Because of the architecture of the 2920 and the fact that several estimates had to be transferred out of each 2920, it was decided to transfer the estimates out of the 2920s serially. A parallel rather than serial interface to the EXORcisor was chosen in an attempt to maintain as universal as possible an interface. Serial in, parallel out shift registers (74164) were used to construct this interface. Because a submultiple sampling rate was used to effect the required filter sampling rates, a number of effectively wasted program passes were spent between actual filter iterations. These program passes were used to transfer out the bits of the digital spectral estimates. It was decided to pass an 8 bit (twos complement) spectral estimate, and therefore 8 successive program passes were used to shift out the values. The 2920 is capable of generating an end of program pulse (EOP) through the use of the EOP instruction. The EOP pulse was, therefore, used to control the shifting of the registers. Since it was desired to have the shift registers stop shifting after the 8 passes, a control output was produced by the 2920 (through software timers) to enable the shifting for only 8 program passes. Software timers in the 2920 were also used to establish one second intervals, and the operations of the filters were arranged so that no new filter output values were determined within the 8 program pass period. The shift control signal was also used to trigger an in-

interrupt in the EXORcisor, thereby signalling that a read of the estimate values should occur.

5.2.2 Interface Problems

There were a number of problems encountered in developing this interface. The two most serious problems stemmed from glitches which were present in the 2920 output signals. The first of these problems was that the EOP pulse generated by the 2920 contained 2 spikes, each distinct enough to trigger a TTL gate input. This waveform can be seen in Fig. 5.2. The second problem was related to the control signal used to enable the shifting. A digital output of the 2920 was used to produce a logical one, under software control, for the required 8 program pass period. This implies, given the architecture of the 2920, that the output was successively 'rewritten' with each program pass. The successive writing of a logical one with a 2920 is corrupted with a glitch, also large enough to trigger a TTL gate input. This waveform can be seen in Fig. 5.3.

These problems were addressed as follows. The end of program pulse was used to trigger a monostable multivibrator (one-shot) circuit (74LS221) in order to generate the desired single pulse. This circuit was not affected by the second spike of the pulse. The control line was deglitched through the use of a CMOS NOR gate, and a resistor-capacitor pair. A CMOS logic gate input threshold voltage is 50% of its supply voltage. Since the gate draws very little current, a resistor-capacitor pair can be arranged to ensure that one of the gate input lines does not reach the threshold voltage unless a change in the input voltage occurs

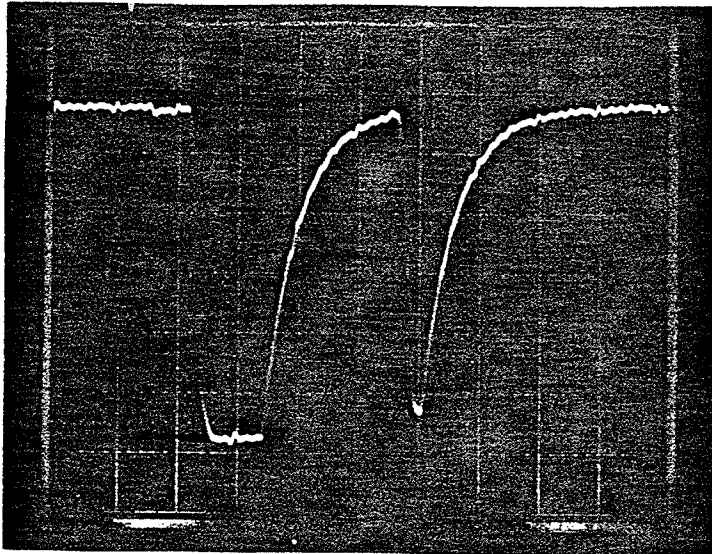


Figure 5.2: 2920 End of program pulse waveform

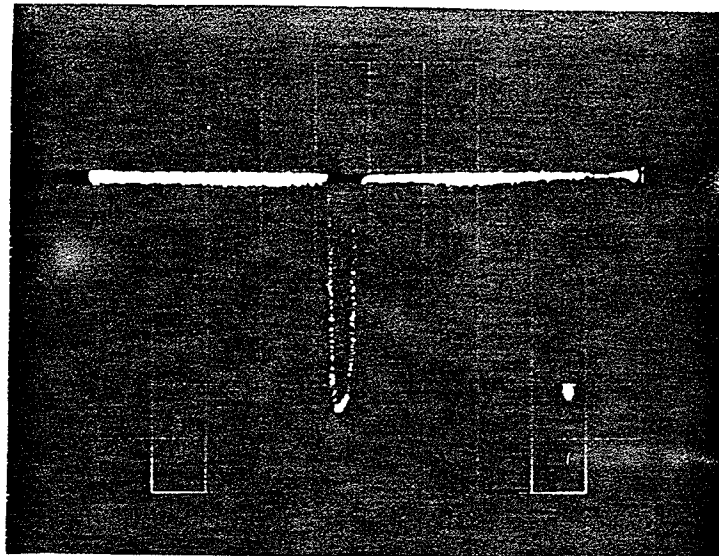


Figure 5.3: 2920 successive logic 1 output waveform

for longer than a selected time. In this way, the control line can be effectively deglitched. Complete circuit diagrams for the 2920/EXORCisor interface and the 2920 programs are included in the appendix.

5.2.3 2920/EXORcisor Synchronization

Since three 2920s were used to accomplish the signal processing, synchronization between the devices and the EXORcisor had to be considered. First of all, it was desired that the 2920s all have their estimates shifted into their shift registers synchronously. In order to accomplish this, one 2920 was chosen as the master. The end of program pin on the 2920 can act as an input or output, and so the EOP generated by the master 2920 (used to control the shift registers) was also used to synchronize the program passes of the 2920s. If the EOP lines of two 2920s are tied together, the devices will be assured to be operating within 6 program steps of each other (the difference arises from the fact that the 2920 prefetches instructions in order to increase its operating rate). On powerup, therefore, the 2920s will all be synchronized after the master 2920 reaches its end of program instruction the first time. In order to synchronize the software timers in each of the 2920s, a start control signal was generated using a CMOS NOR gate and a resistor-capacitor pair. This circuit provided a signal which remained high for several seconds after powerup, and then went low for the remainder of the powered on period. This signal, taken as input by each of the 2920s, provided a means of synchronizing all the software timers. The 2920s all derived their clocks from a common circuit, and therefore all operated

in complete synchronism. The interface and synchronization scheme is general enough to accomodate any number of 2920s, and additional elements can simply be added to the structure in parallel.

5.2.4 EXORcisor Data Acquisition

In order to accommodate the acquisition of the parallel data from the 2920/shift registers, four Motorola 6821 parallel interface adapters (PIAs) were used. This scheme provided parallel access to 8 bytes of input. The outputs of the shift registers were simply connected to the PIAs. A clock was added to the EXORcisor in order to allow correlation of the recorded information to the manually classified sleep record. A Motorola 6840 programmable timer was used to accomplish this function.

As mentioned earlier, the choice of realization structure for the implementation of a digital filter is based on many factors. The form chosen in this study was based on recommendations suggested by Chen [35]. One of the considerations in such a decision is whether the filter structure will always return, from a given state, to a stable zero output for a continuous zero input. For a stable filter design, the structure suggested by Chen assures that the implementation will always be stable, given an initial state of zero in all the filter delays and a bandlimited input signal. Recent studies have shown, however, that these structures may lead to unstable filters given a completely random initial state [39,40]. The memory in the 2920 is in a random state on powerup. Given the architecture of the 2920, it is not possible to effect a zero initial state for all the filter delays without sacrificing many program instruction steps. It was decided, therefore, to implement

the filters without establishing an initial zero state. The possibility of having an unstable structure was realized only late in the study, as only a few of the many possible initial random states of the 2920 lead to unstable conditions. While the recent studies of this instability have suggested structures less susceptible to the problem, they were not implemented in this study. To accomodate the problem, therefore, a program was developed for the EXORcisor to display the 2920 outputs. This routine, which is included in the appendix, allows the user to see the outputs of the 2920s once a second. When no signals are being applied to the 2920s, the outputs should equal zero. After the 2920s are powered on, time is required for the outputs of the filters to stabilize to zero (given the initial random state). Should a structure become unstable, the output will not settle to zero. If this occurs, the 2920s can be powered down for a moment, and a different random state will be established in the 2920 RAM on powerup. In practice, it was found that the filters were almost always stable, and that a single powerdown/up generally cleared any problem which might be present. In order to stop the display process, the user simply enters any key on the keyboard.

5.3 EXORCISOR/AMDAHL INTERFACE

The interface between the EXORcisor and the Amdahl was implemented in the following manner. A program was developed for the EXORcisor to allow it to look like an intelligent terminal to the Amdahl. The user initiates the terminal emulation program (included in the Appendix), and receives a prompt from the program. The user can then logically establish communications between the EXORcisor system terminal and the Amdahl

through the use of an asynchronous communications interface adapter (ACIA) card in the EXORcisor bus. The EXORcisor program passes the messages between the two ports until an escape character is sent from the user's terminal. At that time, the EXORcisor program sends the user a prompt, and allows the user to either go back 'on line', upload an EXORcisor file, or end the procedure.

In order to conduct machine to machine communications, there must not exist the possibility of one machine sending information to the other more quickly than the second can accomodate. Most large, multi-user systems, however, are managed by complex operating systems and activity schedulers, and have difficulty assuring that absolutely no data 'over-runs' will occur with their online users. This can present problems for the user of a microprocessing system who wishes to transfer data to the main-frame. The University of Manitoba Computer Centre has addressed this problem in the following manner. The terminal control program operating in their Amdahl main-frame site has been modified to allow the user to specify a character which is to be sent to the users terminal just before input will be accepted. This feature, when used within a very limited range of main-frame activities, will assure that data over-runs will not occur for a single burst of input of up to 150 characters followed by a carriage return. The user then has to wait until the Amdahl responds with the specified character before the next burst can be sent. The software operating in the Amdahl which is accepting the incoming data, however, must not provide any input prompting (ie. line numbers), or the assurance is lost. Also, the user should disable the message receive capability of the userid in order to assure that nothing

disrupts the data transfer. For the University of Manitoba Amdahl site, this can be accomplished (with the MANTES monitor system) by locating to the file in which the data is to be placed, and entering: 't profile noopmsgs nointercom;t terminal write(xon);i la sup noi'. The data can then be sent in the manner described. This scheme is effected by the terminal emulation routine operating in the EXORcisor. If the EXORcisor waits for the start character from the Amdahl for more than 10 seconds, the routine assumes that there has been a breakdown in communications and terminates the procedure. (In practice it has been found that, when the Amdahl is heavily loaded, this period can be exceeded even though communications have not broken down). The EXORcisor routine provides a line count as the lines are being sent, and logically reconnects the user terminal to the Amdahl when the upload has been completed. (For the MANTES user, the the following should then be entered to reset the user-id: 't terminal nowrite;t profile opmsgs intercom').

5.4 SYSTEM TESTING

The system was tested in the following manner. As mentioned earlier, the filter performances were tested with standard frequency response techniques. The digital transfer scheme was tested with a 2920 and the EXORcisor programmed to simply sample an input signal, transfer the digital sample to the EXORcisor, and store the sample on the EXORcisor disk. Sinusoidal and dc test patterns were captured in this manner. These patterns were then transferred from the EXORcisor to the Amdahl, and were plotted on the Versatec plotter. The actual filter programs with the digital interface scheme were also checked by running the pow-

er-on routine in the EXORcisor, and observing the spectral estimates as passed from the 2920s. Various frequencies and amplitudes of a sinusoidal input were applied in this test.

For evaluation purposes, 2 hours of sleep signals (2 EEG and an EOG) were recorded with a Hewlett Packard 3960 FM instrumentation tape recorder. These signals were also plotted with a polygraphic recorder for manual classification. The beginning of the recorded signal was identified with a synchronization mark to allow correlation with the manually classified sleep record.

5.5 OPERATION OVERVIEW

In order to process a given sleep signal recording, the following procedure is used. Finding the synchronization mark on the tape is accomplished with a Hewlett Packard 7402A strip chart recorder. With time zero of the recording established, the 2920s are powered on and allowed to stabilize. This is checked with the power-on routine, as explained earlier. When the filters are stable, the power-on routine is terminated by entering any key on the system terminal.

Recording of the spectral characteristics is conducted as follows. The user initiates the record program in the EXORcisor (the program listing is included in the Appendix). The record command is appended with the name of an EXORcisor disk file into which the recorded data is to be stored. The routine checks for disk space and opens the file. The routine then prompts the user and waits for a carriage return to synchronize time zero. The user then starts the tape recorder and enters a carriage return at the same time. The real-time clock in the EXORcisor

is zeroed when the carriage return is received, and the interrupt from the 2920s is enabled. The EXORcisor is then interrupted once a second by the 2920s to read the current spectral estimates. The 2920s shift their most current estimates into the shift registers just before the interrupt, and the shift registers remain 'frozen' until just before the next interval arrives. The EXORcisor routine compares the values read with those last recorded, and determines if any of the newly read values differ beyond a threshold from those corresponding in the last record iteration. The difference allowed is adjustable for each of the estimates. If a difference is exceeded, the newly read values are added to the list of recorded values, and appended by the current reading of the clock. These values are then taken as the comparison values. This scheme effects some data compression in that no new entries are added to the list of recorded values if the incoming estimates are the same within the thresholds. The recording process is terminated if either the end of the EXORcisor RAM is reached, or the user enters any key on the system console. At that time, the recorded hexadecimal values are written into the specified disk file. While the recording is in progress, the current number of records written into the list is displayed on the system console.

Before the collected estimates can be transferred to the Amdahl, they must be converted into ASCII character images of the hexadecimal values. It is also desirable to have the times of the recording transitions placed in the frame of reference of the polygraphic recording (for comparison with the manually classified record). The polygraphic recording paper is numbered, with one page corresponding to 30 seconds of record-

ing. The MAP routine, which is run in the EXORcisor, is used to accomplish these functions (program is included in the Appendix). This routine prompts the user for the input hex file name, an output file name, the starting page number of the recording, and the offset of the starting pulse (in seconds) from the start of the page. The recorded values (2s complement 8 bit) are converted to an ASCII image of signed base 10 integers (ranging between -128 and +127), with one line used for each recorded transition. The mapped image of the estimates is then transmitted to the Amdahl with the scheme described earlier.

Chapter VI

RESULTS AND DISCUSSION

The described system was used to estimate the spectrum for a patient exhibiting normal and abnormal sleep waveforms. The sensitivity in detecting changes in the incoming spectral estimates was set at any change beyond the least significant bit of an estimate. With this sensitivity to change, a 90 minute recording was possible before a memory limit of 25 Kbytes of data was reached in the EXORcisor. The record which was chosen provided examples of sleep stages 1 through 4, and of awake sleep. Approximately 60 minutes into the recording, the patient began a period of frequently aroused or disturbed sleep. The EOG channel, although processed and recorded, was not considered in the preliminary evaluation of the recorded estimates. Likewise, the 12 to 14 Hz band estimate used to detect sleep spindles was not considered in this study. The primary concern in this preliminary evaluation was the correlation between the observed spectral estimates in the low and mixed bands, and the manually classified sleep record.

The direct plot of these estimates, as recorded at the end of each one second interval over approximately the first 36 minutes, can be seen in Fig. 6.1. The classifications of the record, as determined manually, have been superimposed on the plot. (The manual classification was done without the clinician seeing the recorded spectral estimates). It can be observed that the spectrum does not appear very stationary over that

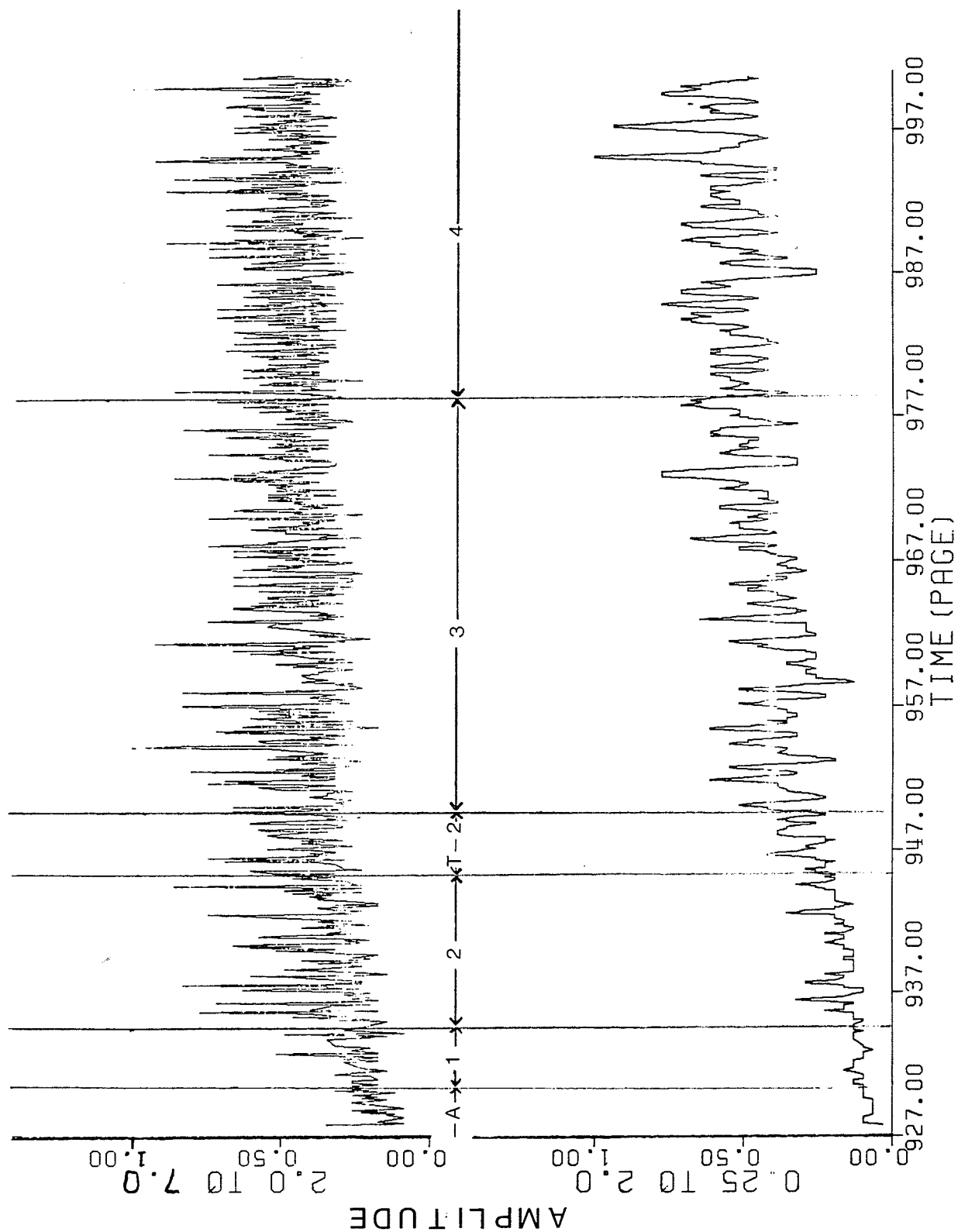


Figure 6.1: Spectral estimates sampled at 1 second intervals

interval, although general trends can be seen to correlate fairly well with the manually classified record. The clinician applies some averaging when manually assessing the plotted sleep signals. It is also clear, from Fig. 6.1, that it would be difficult to establish a pattern recognition scheme able to accommodate the detail presented in a long sleep segment without some averaging. In an attempt to establish what averaging or epoch interval would be most suitable for the classification of normal and abnormal sleep, various epoch rules were applied to the recorded estimates. Plots of the spectral estimates with the application of 5, 10, 30, and 60 second epoch rules can be seen in Figs. 6.2, 6.3, 6.4, and 6.5. The program generating these plots is included in the Appendix. From these plots it can be seen that the spectral estimates become less variable when considered with a longer averaging period. Past studies have commonly used epoch averaging periods of 30 to 60 seconds. These plots show that, with such an averaging period, there is a distinct gross correlation between the spectral estimates and the manually classified sleep stage. The 30 and 60 second epoch rule plots show how past studies have been able to use simple level detectors to establish sleep stage. The exact threshold levels to be applied, however, can vary even within a single recording.

The difficulty in the application of a longer averaging period arises when sleep stage transitions occur in fairly rapid succession. An example of this problem can be seen in Fig. 4.5, where a 10 second epoch rule has been applied to a later portion of the same patient's record. During this period, the patient went through a number of sleep stages in a short time. The correlation between the superimposed manual classifi-

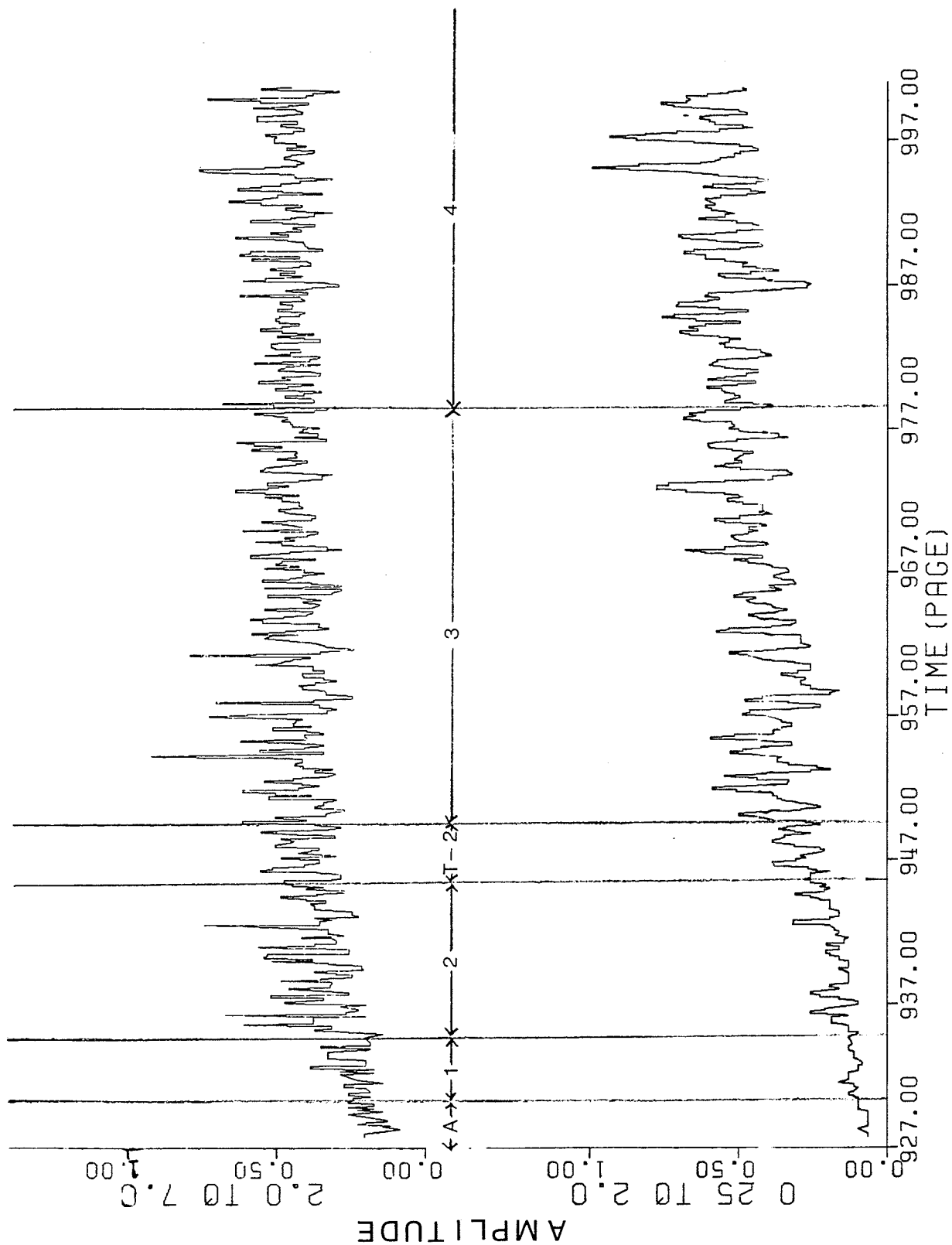


Figure 6.2: Spectral estimates with 5 second epoch rule

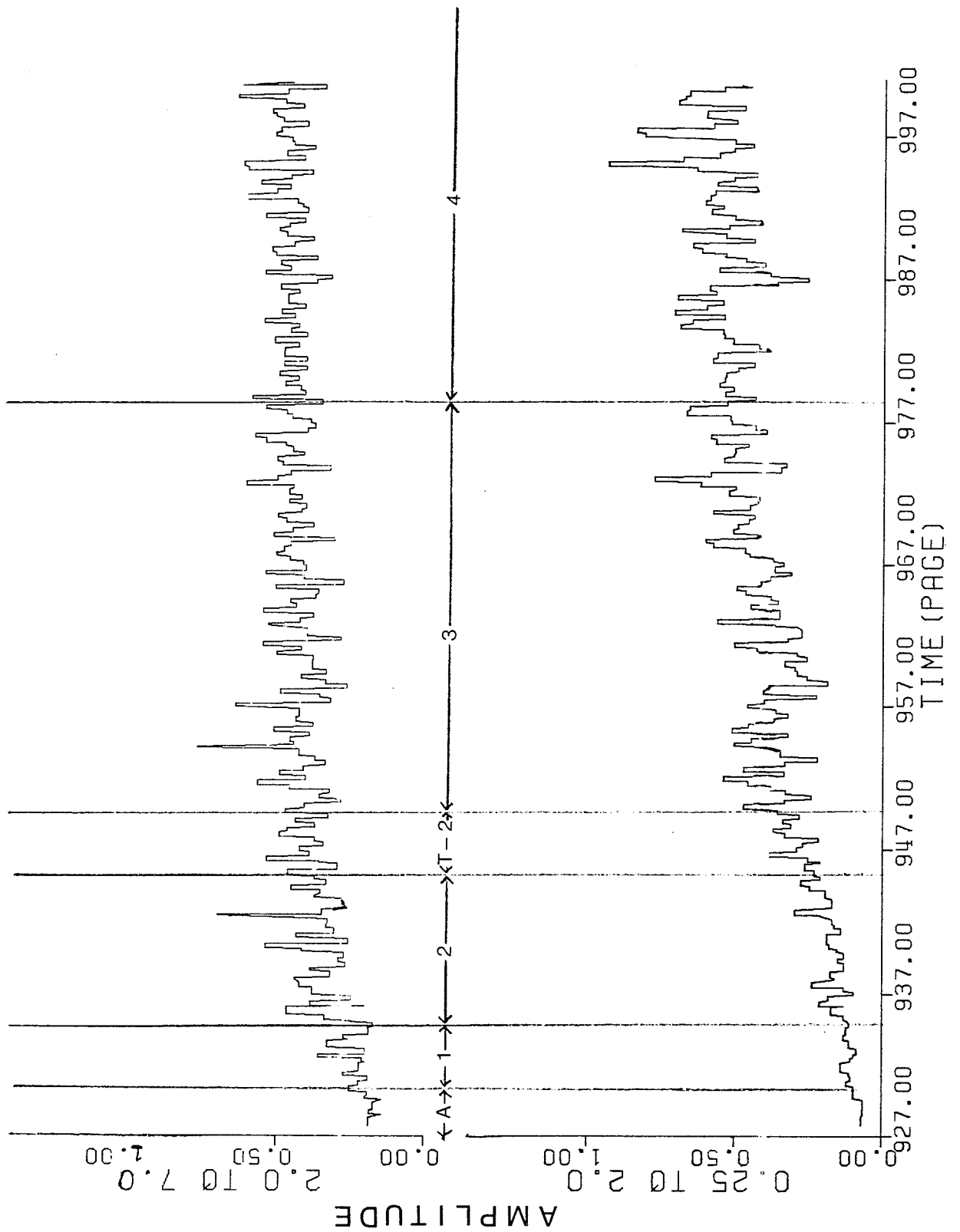


Figure 6.3: Spectral estimates with 10 second epoch rule

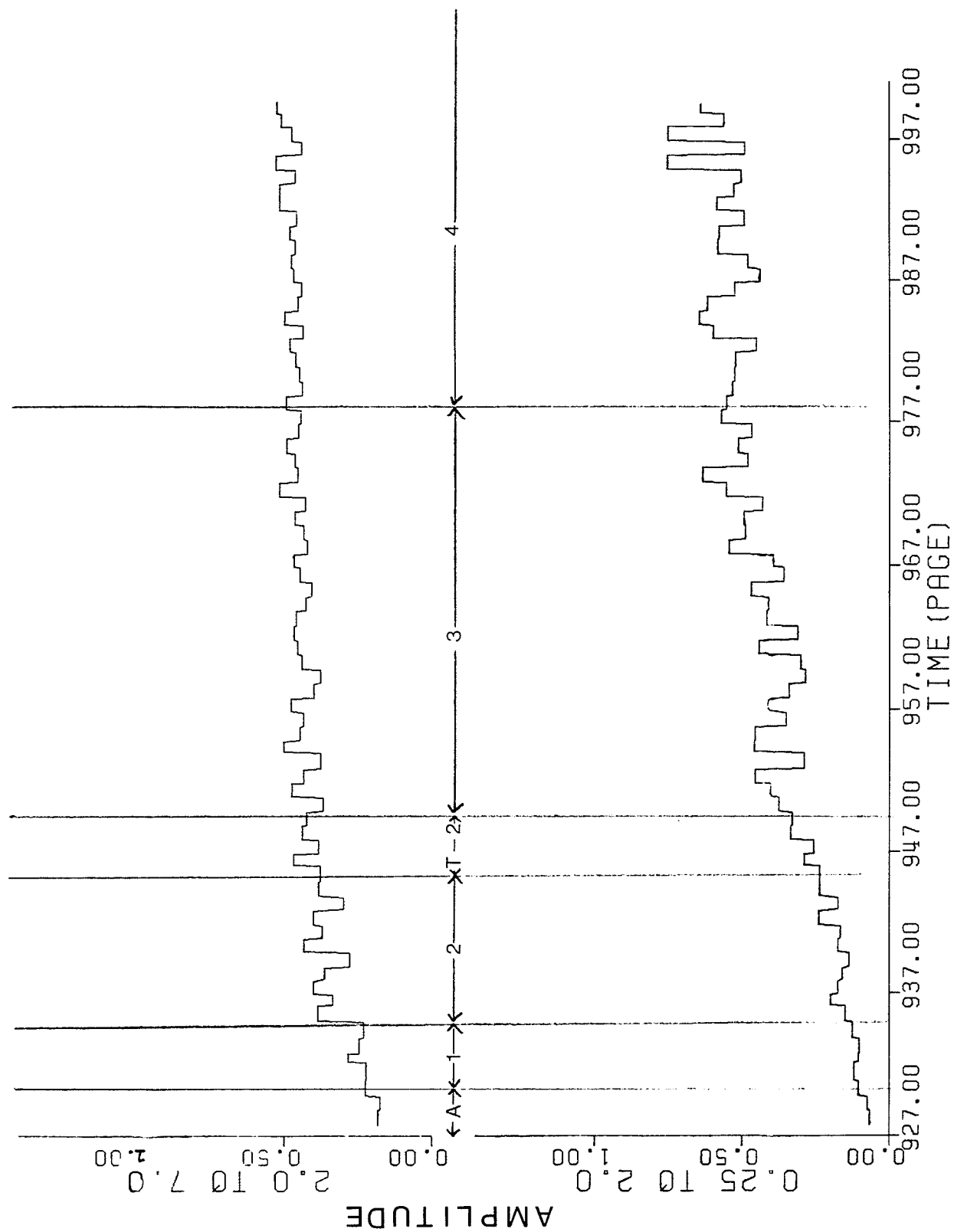


Figure 6.4: Spectral estimates with 30 second epoch rule

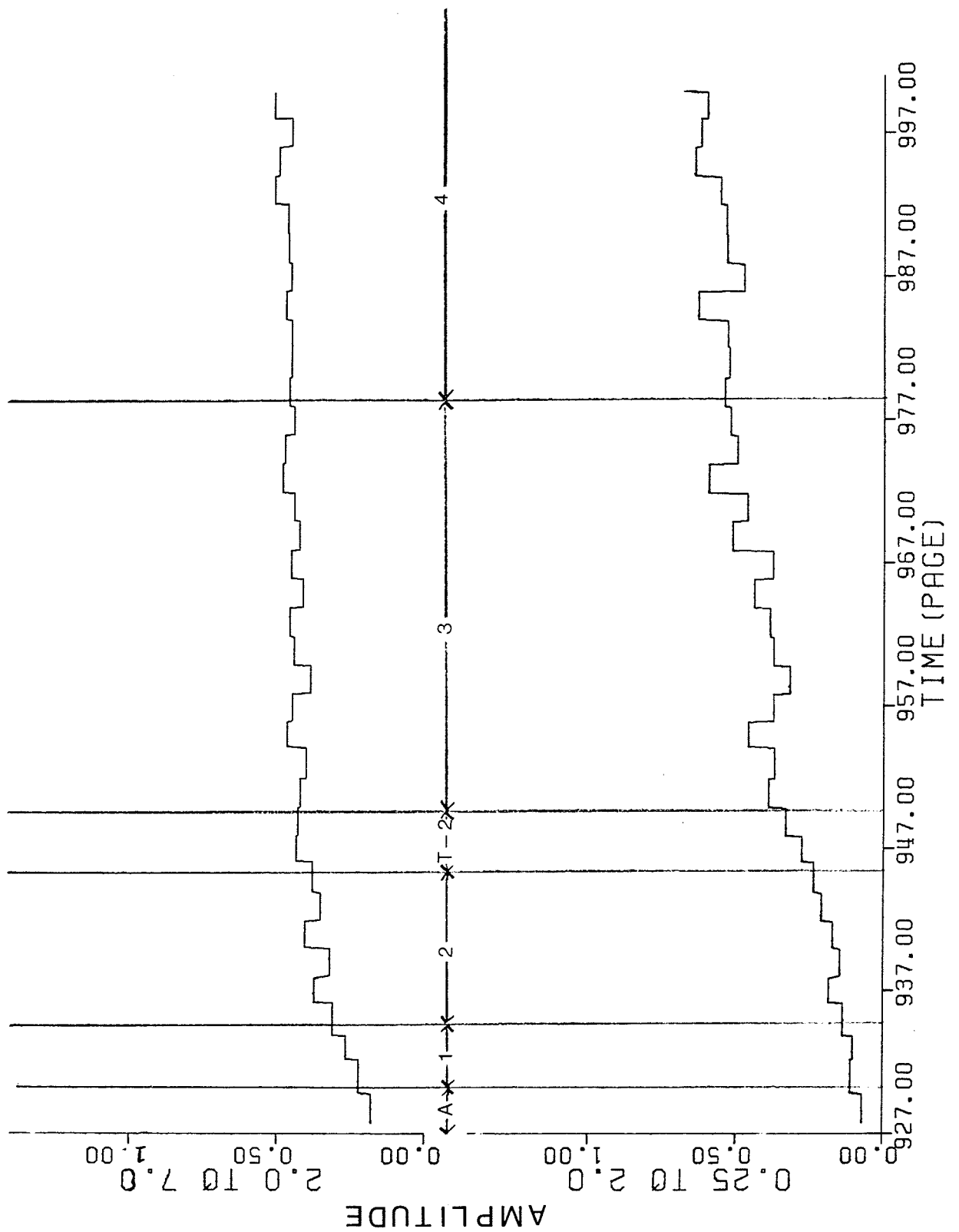


Figure 6.5: Spectral estimates with 60 second epoch rule

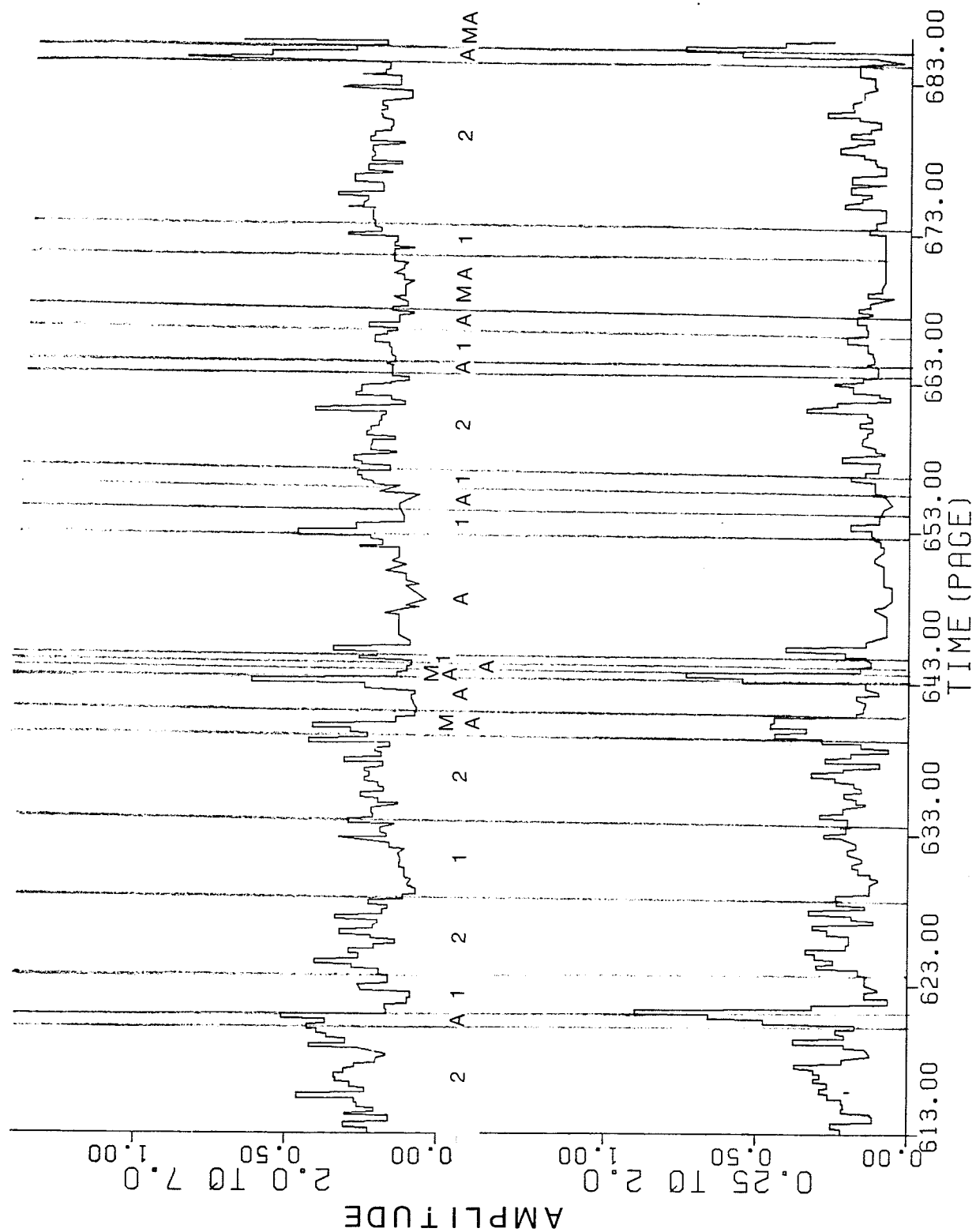


Figure 4.5: Spectral estimates during abnormal sleep (10 sec epoch)

cation and the spectral estimates is not at all clear. Past studies have justified the use of long averaging periods with the fact that normal sleep patterns see changes which occur only over long time intervals. The use of such an averaging rule may affect the ability of a system to identify a brief sleep segment. If a brief averaging rule is used however, the spectral estimates are not stationary. The close dependence on choosing an averaging period of 30 to 60 seconds in order to observe a stationary spectrum has not been reported and was somewhat unexpected.

It appears, therefore, that a tradeoff must occur in the use of an averaging period. For a given pattern recognition scheme, the use of a short averaging period will require the consideration of a large amount of detail in the classification of a normal, lengthy sleep stage interval. The application of a long averaging interval, however, may prevent the identification of a brief sleep stage segment (the averaging implied by the step responses of the filters estimating the spectrum may prevent such an identification). The evaluation of this problem must be integrated into a study considering a specific pattern recognition scheme. A variable averaging technique may be an avenue for consideration.

Chapter VII

CONCLUSIONS AND RECOMMENDATIONS

This study has focused on the development of a computer-aided method for sleep signal classification, and has considered the use of inexpensive microprocessing hardware for the task of sleep signal spectral estimation. It was found that, for normal sleep, the spectral estimates evaluated by the developed system correlated well with the manually classified sleep record. These estimates (as sampled once a second) were found to be non-stationary within the manually classified sleep stage segments, although a gross correlation between the estimates and the manual classification was still apparent. The application of averaging to the spectral estimates was examined, and it was found that the averaged spectral estimates became progressively more stationary with increasing averaging period. This averaging simplified the correlation between the manual classifications and the spectral estimates. An abnormal sleep record was also examined, and it was found that the application of even a brief averaging period could blur the correlation between the manual classification and the spectral estimates. This difficulty arose because disturbed sleep can present frequent sleep stage transitions. It appears, therefore, that a tradeoff must occur in the use of such an averaging or epoch rule.

As a result of this work, it is recommended that the use of syntactic pattern recognition for sleep signal classification be evaluated in a

scheme using the clinician as feedback to the system. This study will have to evaluate, for the specific recognition scheme, the use of a spectral averaging period. The use of a variable averaging period may be an avenue for consideration. Recent advances in pattern recognition, coupled with the approach developed in this study, may lead to an inexpensive scheme to aid in the classification of normal and abnormal sleep. In order to conduct this study, a clinical evaluation of the scheme will be required. It is strongly suggested that a library of recorded and manually classified sleep records be established for this purpose. Such a library will be invaluable to any further study of this subject.

Beyond the specific application considered, the developed method represents a general approach which should be applicable to a number of problems involving extended duration signal characterization and classification.

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Appendix A
FILTER DERIVATIONS

BANDPASS FILTER

ω_1 = upper cutoff frequency [Hz]
 ω_2 = lower cutoff frequency [Hz]
 f = sampling frequency [Hz]

$$h = \frac{\cos[(\omega_1 + \omega_2) 2\pi/2(f)]}{\cos[(\omega_1 - \omega_2) 2\pi/2(f)]} \quad (A.1)$$

$$\bar{\omega} = \frac{h - \cos[\omega_1(2\pi)/f]}{\sin[\omega_1(2\pi)/f]} \quad (A.2)$$

Design a Butterworth Lowpass (2nd order) with cutoff at $\bar{\omega}$

$$H(s) = \frac{K}{s^2 + \sqrt{2}s + 1} \quad \begin{array}{l} \text{second order Butterworth with cutoff} \\ \text{normalized} \end{array} \quad (A.3)$$

$$H(s) = \frac{K'}{s^2 + \sqrt{2}\bar{\omega}s + \bar{\omega}^2} \quad \begin{array}{l} \text{second order Butterworth L.P. with cutoff} \\ \text{at } \bar{\omega} \end{array} \quad (A.4)$$

$$= \frac{K'}{(s-m_1)(s-m_2)} \quad (A.5)$$

$$\text{where: } m_1, m_2 = \frac{-(\sqrt{2}\bar{\omega}) \pm \sqrt{(\sqrt{2}\bar{\omega})^2 - 4\bar{\omega}^2}}{2} \quad (A.6)$$

TRANSFORMATION: Analog lowpass to digital bandpass

$$S = \frac{z^2 - 2hz + 1}{z^2 - 1} \quad (A.7)$$

$$H(z) = \frac{K}{\left(\frac{z^2 - 2hz + 1}{z^2 - 1} - m1\right)\left(\frac{z^2 - 2hz + 1}{z^2 - 1} - m2\right)} \quad (A.8)$$

$$H(z) = \frac{K(z^2 - 1)^2}{[(z^2 - 2hz + 1) - m1(z^2 - 1)] \cdot [(z^2 - 2hz + 1) - m2(z^2 - 1)]} \quad (A.9)$$

$$= \frac{K(z^2 - 1)^2}{[(1 - m1)z^2 - 2h + (m1 + 1)] \cdot [(1 - m2)z^2 - 2hz + (m2 + 1)]} \quad (A.10)$$

$$= \frac{K(z^2 - 1)^2 \left(\frac{1}{1 - m1}\right) \left(\frac{1}{1 - m2}\right)}{\left[z^2 - \left(\frac{2h}{1 - m1}\right)z + \left(\frac{1 + m1}{1 - m1}\right)\right] \cdot \left[z^2 - \left(\frac{2h}{1 - m2}\right)z + \left(\frac{1 + m2}{1 - m2}\right)\right]} \quad (A.11)$$

$$= \frac{K'(z - 1)(z + 1)(z - 1)(z + 1)}{(z - \alpha_1)(z - \alpha_2)(z - \alpha_3)(z - \alpha_4)} \quad (A.12)$$

$$\text{where: } \alpha_1, \alpha_2 = \frac{\left(\frac{-2h}{1 - m1}\right) \pm \sqrt{\left(\frac{2h}{1 - m1}\right)^2 - 4\left(\frac{1 + m1}{1 - m1}\right)}}{2} \quad (A.13)$$

$$\alpha_3, \alpha_4 = \frac{\left(\frac{-2h}{1 - m2}\right) \pm \sqrt{\left(\frac{2h}{1 - m2}\right)^2 - 4\left(\frac{1 + m2}{1 - m2}\right)}}{2} \quad (A.14)$$

Poles α_1 & α_3 and α_2 & α_4 are complex conjugates

BANDPASS FILTER TRANSFER FUNCTION:

$$H(z) = \frac{K(z-1)(z+1)}{z^2 - (\alpha_1 + \alpha_3)z + \alpha_1 \alpha_3} \cdot \frac{(z-1)(z+1)}{z^2 - (\alpha_2 + \alpha_4)z + \alpha_2 \alpha_4} \quad (A.15)$$

LOW PASS FILTER^R

cutoff frequency = ω_p [Hz]

sampling frequency = f_s [Hz]

$$\bar{\omega}_p = \tan \frac{\omega_p(2\pi)}{2(f_s)} = x \quad (A.16)$$

$$\bar{H}(s) = \frac{1}{\left(\frac{s}{x}\right)^2 + \frac{1.414}{x}s + 1} = \frac{x^2}{s^2 + \sqrt{2} x s + x^2} \quad (A.17)$$

$$\text{BILINEAR TRANSFORM: } s = \frac{z-1}{z+1} \quad (A.18)$$

$$= \frac{x^2}{\left(\frac{z-1}{z+1}\right)^2 + \sqrt{2} x \left(\frac{z-1}{z+1}\right) + x^2} \quad (A.19)$$

$$= \frac{x^2(z+1)^2}{(z-1)^2 + \sqrt{2} x (z-1)(z+1) + x^2(z+1)^2} \quad (A.20)$$

$$= \frac{x^2(z+1)(z+1)}{(z-1)(z-1) + \sqrt{2} x (z-1)(z+1) + x^2(z+1)(z+1)} \quad (A.21)$$

$$= \frac{x^2 (z^2 + 2z + 1)}{(z^2 - 2z + 1) + \sqrt{2} x (z^2 - 1) + x^2 (z^2 + 2z + 1)} \quad (\text{A.22})$$

$$= \frac{x^2 (z^2 + 2z + 1)}{(x^2 + \sqrt{2} x + 1)z^2 + (2x^2 - 2)z + (x^2 - \sqrt{2} x + 1)} \quad (\text{A.23})$$

$$= \frac{\frac{x^2}{(x^2 + \sqrt{2} x + 1)} \cdot (z^2 + 2z + 1)}{z^2 + \frac{2x^2 - 2}{(x^2 + \sqrt{2} x + 1)} z + \frac{x^2 - \sqrt{2} x + 1}{x^2 + \sqrt{2} x + 1}} \quad (\text{A.24})$$

$$= \frac{K(z^2 + 2z + 1)}{z^2 + \alpha z + \beta} \quad (\text{A.25})$$

$$\text{where: } \alpha = \frac{2x^2 - 2}{x^2 + \sqrt{2} x + 1} \quad (\text{A.26})$$

$$\beta = \frac{x^2 - \sqrt{2} x + 1}{x^2 + \sqrt{2} x + 1} \quad (\text{A.27})$$

Appendix B
SOFTWARE LISTINGS

```

*JOB  WATFIV  REIMER, NOEXT
C
C *****
C *
C * PROGRAM TO CALCULATE BANDPASS AND LOWPASS FILTER COEFFICIENTS *
C * AND EXPERIMENTALLY EVALUATE SCALING FACTORS TO THE NEAREST *
C * POWER OF 2 TO AVOID SATURATIONS IN THE FILTER CALCULATIONS *
C *
C *****
C
1  REAL W1, W2, F, WP, FS, A, B, C, D, ALP, BET, FIN, K1, K2, K3
2  INTEGER N, I
3  READ, N
4  DO 999 I=1, N
5      READ, W2, W1, F, WP, FS, FIN
6      CALL BAND(W1, W2, F, A, B, C, D)
7      CALL LOW(WP, FS, ALP, BET)
8      CALL SCALE(A, B, C, D, F, ALP, BET, FS, FIN, K1, K2, K3, K4)
9      PRINT 401, IFIX(K1)
10     PRINT 402, IFIX(K2)
11     PRINT 403, IFIX(K3)
12     PRINT 404, K4
13     FORMAT(' ', 'K1=2**', I3)
14     FORMAT(' ', 'K2=2**', I3)
15     FORMAT(' ', 'K3=2**', I3)
16     FORMAT(' ', 'LARGEST OBSERVED OUTPUT = ', F9.7)
17 999  CONTINUE
18      STOP
19  END

C
C *****
C *
C * PROGRAM TO COMPUTE THE COEFFICIENTS FOR A FOURTH ORDER *
C * BUTTERWORTH DIGITAL BANDPASS FILTER USING THE BILINEAR *
C * ANALOG LOWPASS TO DIGITAL BANDPASS METHOD. *
C *
C *****
C
20  SUBROUTINE BAND(W1, W2, F, A, B, C, D)
21  COMPLEX M1, M2, TEMP2, TEMP3, ALP1, ALP2, ALP3, ALP4
22  REAL W1, W2, DIG, H, PI, F, TEMP0, TEMP1, A, B, C, D
23  PRINT 100
24 100  FORMAT('1')
25  PRINT, 'BANDPASS FILTER'
26  PRINT 101, W2
27 101  FORMAT(' ', 'LOWER CUTOFF FREQUENCY = ', F9.5)
28  PRINT 102, W1
29 102  FORMAT(' ', 'UPPER CUTOFF FREQUENCY = ', F9.5)
30  PRINT 103, F
31 103  FORMAT(' ', 'SAMPLING FREQUENCY = ', F6.1)
32  PI=3.141592654
33  TEMP0=((W1+W2)*2*PI)/(2*F)
34  TEMP1=((W1-W2)*2*PI)/(2*F)
35  H=(COS(TEMP0))/(COS(TEMP1))
36  TEMP0=(W1*2*PI)/F
37  DIG=(H-(COS(TEMP0)))/(SIN(TEMP0))
38  TEMP0=(SQRT(2.0))*DIG
39  TEMP1=(TEMP0**2)-(4*(DIG**2))
40  TEMP3=CMPLX(TEMP1, 0.0)
41  M1=(((-TEMP0)+CSQRT(TEMP3))/2.0
42  M2=(((-TEMP0)-CSQRT(TEMP3))/2.0
43  TEMP2=(2*H)/(1-M1)
44  TEMP3=(TEMP2**2)-(4*((1+M1)/(1-M1)))
45  ALP1=(TEMP2+CSQRT(TEMP3))/2.0
46  ALP2=(TEMP2-CSQRT(TEMP3))/2.0
47  TEMP2=(2*H)/(1-M2)
48  TEMP3=(TEMP2**2)-(4*((1+M2)/(1-M2)))
49  ALP3=(TEMP2+CSQRT(TEMP3))/2.0
50  ALP4=(TEMP2-CSQRT(TEMP3))/2.0
51  TEMP0=REAL(ALP1)
52  TEMP1=AIMAG(ALP1)
53  A=-((2*TEMP0)
54  B=(TEMP0**2)+(TEMP1**2)
55  PRINT 104, A
56 104  FORMAT('0', 'A = ', F10.7)
57  TEMP9=-A
58  CALL SIGN(TEMP9)
59  PRINT 105, B
60 105  FORMAT('0', 'B = ', F10.7)
61  TEMP9=B
62  CALL SIGN(TEMP9)
63  TEMP0=REAL(ALP2)
64  TEMP1=AIMAG(ALP2)
65  C=-((2*TEMP0)
66  D=(TEMP0**2)+(TEMP1**2)
67  PRINT 106, C
68 106  FORMAT('0', 'C = ', F10.7)
69  TEMP9=-C
70  CALL SIGN(TEMP9)
71  PRINT 107, D
72 107  FORMAT('0', 'D = ', F10.7)
73  TEMP9=D
74  CALL SIGN(TEMP9)
75  RETURN
76  END

```

```

C *****
C *
C * SUBROUTINE TO LIST THE SIGN DIGIT CANONIC FORM OF A NUMBER *
C *
C *****
C
77      SUBROUTINE SIGN(A)
78      REAL*8 TABLE(16),DIFF,DIFF1,DIFF2
79      INTEGER I,J,NUM(16)
80      PRINT 9, A
81      9      FORMAT('0', 'SIGN DIGIT BREAKDOWN OF ',F11.7)
82      NUM(1)=2
83      TABLE(1)=4.0
84      I=1
85      DO 1 J=2,16
86          TABLE(J)=TABLE(J-1)*0.5
87          NUM(J)=I
88          I=I-1
89      1      CONTINUE
90      DIFF=A
91      WHILE(DABS(DIFF) .GT. .0001) DO
92          I=16
93          WHILE(TABLE(I) .LT. DABS(DIFF)) DO
94              I=I-1
95          END WHILE
96          IF(I .EQ. 16) GO TO 2
97          DIFF1=DABS(TABLE(I) - DABS(DIFF))
98          DIFF2=DABS(TABLE(I+1) - DABS(DIFF))
99          IF(DIFF2 .LT. DIFF1) I=I+1
100      2      IF(DIFF .GT. 0) THEN DO
101          PRINT 10,NUM(I)
102          10      FORMAT(' ',5X,'ADD 2 TO THE POWER ',I3)
103          DIFF=DIFF-TABLE(I)
104      ELSE DO
105          PRINT 30,NUM(I)
106          30      FORMAT(' ',5X,'SUBTRACT 2 TO THE POWER ',I3)
107          DIFF=TABLE(I)+DIFF
108      END IF
109      END WHILE
110      RETURN
111      END

C *****
C *
C * PROGRAM TO CALCULATE THE COEFFICIENTS FOR A SECOND ORDER *
C * BUTTERWORTH LOWPASS SECOND ORDER SECTION *
C *
C *****
C
112      SUBROUTINE LOW(WP,FS,ALP,BET)
113      REAL X,ALP,BET,PI,FS,WP,TEMP
114      PRINT, ' '
115      PRINT, 'LOWPASS FILTER'
116      PRINT 200, WP
117      200      FORMAT(' ', 'BREAK FREQUENCY = ',F10.4)
118      PRINT 201, FS
119      201      FORMAT(' ', 'SAMPLE FREQUENCY = ',F10.4)
120      PI=3.14159
121      X=(WP*2*PI)/(2*FS)
122      X=TAN(X)
123      TEMP=1+((1.414*X)+(X**2))
124      ALP=((2*(X**2))-2)/TEMP
125      PRINT 202,ALP
126      202      FORMAT('0', 'ALPHA = ',F10.7)
127      TEMP1=-ALP
128      CALL SIGN(TEMP1)
129      BET=((1-(1.414*X)+(X**2))/TEMP)
130      PRINT 203,BET
131      203      FORMAT('0', 'BETA = ',F10.7)
132      TEMP1=BET
133      CALL SIGN(TEMP1)
134      RETURN
135      END

C *****
C *
C * PROGRAM TO EXPERIMENTALLY EVALUATE THE SCALING FACTORS *
C * REQUIRED TO AVOID OVERFLOWS (SATURATIONS) IN THE BANDPASS *
C * LOWPASS FILTER SETS. THE SCALING FACTORS ARE EVALUATED TO THE *
C * NEAREST POWER OF 2 WHICH RESULTS IN NO SATURATION. *
C *
C *****
C
136      SUBROUTINE SCALE(A,B,C,D,FREQ,ALP,BET,FS,FREQIN,
137      *K1,K2,K3,K4)
138      REAL A,B,C,D,D1,D2,D3,D4,FREQIN,FREQ,X
139      REAL INCREM,K1,K2,K3,K4,DX,DY,ALP,BET
140      INTEGER NMAX,N,I,CNTL,LAST,CNTL1,NUM,NUMEND
141      K1=1
142      K2=1
143      K3=1
144      NMAX=IFIX(FREQ/FREQIN)
145      INCREM=(2*3.14159)/NMAX
146      NUMEND=IFIX(FREQ/FS)
147      CNTL1=1
148      1      N=0
149      X=0
150      NUM=0
151      D1=0

```

```

151      D2=0
152      D3=0
153      D4=0
154      DX=0
155      DY=0
156      CNTL=0
157      LAST=10*NMAX
158      K4=0
159      DO 10 I=1, LAST
160      CALL CYCLE(CNTL, A, B, C, D, K1, K2, K3, D1, D2, D3, D4,
161      * N, X, NMAX, INCREM, NUM, NUMEND, DX, DY, ALP, BET, K4)
162      CONTINUE
163      CNTL=CNTL1
164      K4=0
165      DO 20 I=1, LAST
166      CALL CYCLE(CNTL, A, B, C, D, K1, K2, K3, D1, D2, D3, D4,
167      * N, X, NMAX, INCREM, NUM, NUMEND, DX, DY, ALP, BET, K4)
168      IF (CNTL .EQ. 99) GO TO 99
169      IF (CNTL .EQ. 98) GO TO 98
170      IF (CNTL .EQ. 97) GO TO 97
171      CONTINUE
172      CNTL1=CNTL1+1
173      IF (CNTL1 .EQ. 4) RETURN
174      GO TO 1
175      99      K1=K1-1
176      IF (K1 .EQ. -14) THEN DO
177      PRINT, 'K1 SCALING POWER LESS THAN 2**-14'
178      GO TO 96
179      END IF
180      GO TO 1
181      98      K2=K2-1
182      IF (K2 .EQ. -14) THEN DO
183      PRINT, 'K2 SCALING POWER LESS THAN 2**-14'
184      GO TO 96
185      END IF
186      GO TO 1
187      97      K3=K3-1
188      IF (K3 .EQ. -14) THEN DO
189      PRINT, 'K3 SCALING POWER LESS THAN 2**-14'
190      GO TO 96
191      END IF
192      GO TO 1
193      96      PRINT, 'SCALING ABORTED'
194      RETURN
195      END

```

C *****
C *
C * SUBROUTINE TO SIMULATE A DELAY ITERATION OF THE DIGITAL FILTERS*
C *
C *****
C

```

194      SUBROUTINE CYCLE(CNTL, A, B, C, D, K1, K2, K3, D1, D2, D3, D4,
195      * N, X, NMAX, INCREM, NUM, NUMEND, DX, DY, ALP, BET, K4)
196      INTEGER CNTL, NMAX, N, NUM, NUMEND
197      REAL A, B, C, D, VAL, K1, K2, K3, K4, D1, D2, D3, D4, X, INCREM
198      REAL TEMP1, TEMP2, TEMP3, TEMP4, TEMP5, DX, DY, ALP, BET
199      VAL=SIN(X)
200      N=N+1
201      X=X+INCREM
202      IF (N .EQ. NMAX) THEN DO
203      N=0
204      X=0
205      END IF
206      TEMP1=D1*A
207      IF (TEMP1 .GT. 1) TEMP1=1
208      IF (TEMP1 .LT. -1) TEMP1=-1
209      TEMP2=D2*B
210      IF (TEMP2 .GT. 1) TEMP2=1
211      IF (TEMP2 .LT. -1) TEMP2=-1
212      TEMP3=(-TEMP1)+(-TEMP2)
213      IF (TEMP3 .GT. 1) TEMP3=1
214      IF (TEMP3 .LT. -1) TEMP3=-1
215      TEMP4=(VAL*(2.0**K1))+TEMP3
216      IF (TEMP4 .GT. 1) TEMP4=1
217      IF (TEMP4 .LT. -1) TEMP4=-1
218      TEMP5=TEMP4-D2
219      IF (TEMP5 .GT. 1) TEMP5=1
220      IF (TEMP5 .LT. -1) TEMP5=-1
221      IF (CNTL .EQ. 1) THEN DO
222      IF (ABS(TEMP1) .GE. 1) CNTL=99
223      IF (ABS(TEMP2) .GE. 1) CNTL=99
224      IF (ABS(TEMP3) .GE. 1) CNTL=99
225      IF (ABS(TEMP4) .GE. 1) CNTL=99
226      IF (ABS(TEMP5) .GE. 1) CNTL=99
227      END IF
228      D2=D1
229      D1=TEMP4
230      TEMP1=D3*C
231      IF (TEMP1 .GT. 1) TEMP1=1
232      IF (TEMP1 .LT. -1) TEMP1=-1
233      TEMP2=D4*D
234      IF (TEMP2 .GT. 1) TEMP2=1
235      IF (TEMP2 .LT. -1) TEMP2=-1
236      TEMP3=(-TEMP1)+(-TEMP2)
237      IF (TEMP3 .GT. 1) TEMP3=1
238      IF (TEMP3 .LT. -1) TEMP3=-1
239      TEMP4=(TEMP5*(2.0**K2))+TEMP3
240      IF (TEMP4 .GT. 1) TEMP4=1
241      IF (TEMP4 .LT. -1) TEMP4=-1

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241      TEMP5=TEMP4-D4
242      IF (TEMP5 .GT. 1 ) TEMP5=1
243      IF (TEMP5 .LT. -1) TEMP5=-1
244      IF (CNTL .EQ. 2) THEN DO
245          IF (ABS(TEMP1) .GE. 1) CNTL=98
246          IF (ABS(TEMP2) .GE. 1) CNTL=98
247          IF (ABS(TEMP3) .GE. 1) CNTL=98
248          IF (ABS(TEMP4) .GE. 1) CNTL=98
249          IF (ABS(TEMP5) .GE. 1) CNTL=98
250      END IF
251      D4=D3
252      D3=TEMP4
253      NUM=NUM+1
254      IF (NUM .EQ. NUMEND) THEN DO
255          NUM=0
256          TEMP1=ALP*DX
257          IF (TEMP1 .GT. 1 ) TEMP1=1
258          IF (TEMP1 .LT. -1) TEMP1=-1
259          TEMP2=BET*DY
260          IF (TEMP2 .GT. 1 ) TEMP2=1
261          IF (TEMP2 .LT. -1) TEMP2=-1
262          TEMP3=(-TEMP1)+(-TEMP2)
263          IF (TEMP3 .GT. 1 ) TEMP3=1
264          IF (TEMP3 .LT. -1) TEMP3=-1
265          TEMP5=ABS(TEMP5)
266          TEMP4=(TEMP5*(2.0**K3))+TEMP3
267          IF (TEMP4 .GT. 1 ) TEMP4=1
268          IF (TEMP4 .LT. -1) TEMP4=-1
269          TEMP5=(DX*2)+DY+TEMP4
270          IF (TEMP5 .GT. 1 ) TEMP5=1
271          IF (TEMP5 .LT. -1) TEMP5=-1
272          DY=DX
273          DX=TEMP4
274          IF (CNTL .EQ. 3) THEN DO
275              IF (ABS(TEMP1) .GE. 1) CNTL=97
276              IF (ABS(TEMP2) .GE. 1) CNTL=97
277              IF (ABS(TEMP3) .GE. 1) CNTL=97
278              IF (ABS(TEMP4) .GE. 1) CNTL=97
279              IF (ABS(TEMP5) .GE. 1) CNTL=97
280          END IF
281          IF (K4 .LT. TEMP5) K4=TEMP5
282      END IF
283      RETURN
284  END
$ENTRY

```

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BANDPASS FILTER
LOWER CUTOFF FREQUENCY = 0.25000
UPPER CUTOFF FREQUENCY = 2.00000
SAMPLING FREQUENCY = 50.0
A = -1.9585290
SIGN DIGIT BREAKDOWN OF 1.9585290
ADD 2 TO THE POWER 1
SUBTRACT 2 TO THE POWER -5
SUBTRACT 2 TO THE POWER -7
SUBTRACT 2 TO THE POWER -9
SUBTRACT 2 TO THE POWER -11
B = 0.9597065
SIGN DIGIT BREAKDOWN OF 0.9597065
ADD 2 TO THE POWER 0
SUBTRACT 2 TO THE POWER -5
SUBTRACT 2 TO THE POWER -7
SUBTRACT 2 TO THE POWER -10
SUBTRACT 2 TO THE POWER -12
C = -1.7178450
SIGN DIGIT BREAKDOWN OF 1.7178450
ADD 2 TO THE POWER 1
SUBTRACT 2 TO THE POWER -2
SUBTRACT 2 TO THE POWER -5
SUBTRACT 2 TO THE POWER -10
D = 0.7634901
SIGN DIGIT BREAKDOWN OF 0.7634901
ADD 2 TO THE POWER 0
SUBTRACT 2 TO THE POWER -2
ADD 2 TO THE POWER -6
SUBTRACT 2 TO THE POWER -9
SUBTRACT 2 TO THE POWER -13

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LOWPASS FILTER
BREAK FREQUENCY = 0.0250
SAMPLE FREQUENCY = 12.5000
ALPHA = -1.9822320
SIGN DIGIT BREAKDOWN OF 1.9822320
ADD 2 TO THE POWER 1
SUBTRACT 2 TO THE POWER -6
SUBTRACT 2 TO THE POWER -9
SUBTRACT 2 TO THE POWER -12
BETA = 0.9823886
SIGN DIGIT BREAKDOWN OF 0.9823886
ADD 2 TO THE POWER 0
SUBTRACT 2 TO THE POWER -6
SUBTRACT 2 TO THE POWER -9
K1=2** -9
K2=2** -1
K3=2** -11

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LARGEST OBSERVED OUTPUT = 0.6824680
 BANDPASS FILTER
 LOWER CUTOFF FREQUENCY = 2.00000
 UPPER CUTOFF FREQUENCY = 7.00000
 SAMPLING FREQUENCY = 50.0
 A = -1.7093410
 SIGN DIGIT BREAKDOWN OF 1.7093410
 ADD 2 TO THE POWER 1
 SUBTRACT 2 TO THE POWER -2
 SUBTRACT 2 TO THE POWER -5
 SUBTRACT 2 TO THE POWER -7
 SUBTRACT 2 TO THE POWER -9
 ADD 2 TO THE POWER -12
 ADD 2 TO THE POWER -13
 B = 0.7812762
 SIGN DIGIT BREAKDOWN OF 0.7812762
 ADD 2 TO THE POWER 0
 SUBTRACT 2 TO THE POWER -2
 ADD 2 TO THE POWER -5
 C = -1.0809290
 SIGN DIGIT BREAKDOWN OF 1.0809290
 ADD 2 TO THE POWER 0
 ADD 2 TO THE POWER -4
 ADD 2 TO THE POWER -6
 ADD 2 TO THE POWER -9
 ADD 2 TO THE POWER -10
 SUBTRACT 2 TO THE POWER -13
 D = 0.5283684
 SIGN DIGIT BREAKDOWN OF 0.5283684
 ADD 2 TO THE POWER -1
 ADD 2 TO THE POWER -5
 SUBTRACT 2 TO THE POWER -9
 SUBTRACT 2 TO THE POWER -10

LOWPASS FILTER
 BREAK FREQUENCY = 0.2000
 SAMPLE FREQUENCY = 50.0000
 ALPHA = -1.9644670
 SIGN DIGIT BREAKDOWN OF 1.9644670
 ADD 2 TO THE POWER 1
 SUBTRACT 2 TO THE POWER -5
 SUBTRACT 2 TO THE POWER -8
 SUBTRACT 2 TO THE POWER -11
 ADD 2 TO THE POWER -13
 BETA = 0.9650878
 SIGN DIGIT BREAKDOWN OF 0.9650878
 ADD 2 TO THE POWER 0
 SUBTRACT 2 TO THE POWER -5
 SUBTRACT 2 TO THE POWER -8
 ADD 2 TO THE POWER -12

K1=2** -4
 K2=2** 0
 K3=2** -12
 LARGEST OBSERVED OUTPUT = 0.9679918
 BANDPASS FILTER
 LOWER CUTOFF FREQUENCY = 12.00000
 UPPER CUTOFF FREQUENCY = 14.00000
 SAMPLING FREQUENCY = 200.0
 A = -1.8149230
 SIGN DIGIT BREAKDOWN OF 1.8149230
 ADD 2 TO THE POWER 1
 SUBTRACT 2 TO THE POWER -3
 SUBTRACT 2 TO THE POWER -4
 ADD 2 TO THE POWER -9
 ADD 2 TO THE POWER -11
 B = 0.9587363
 SIGN DIGIT BREAKDOWN OF 0.9587363
 ADD 2 TO THE POWER 0
 SUBTRACT 2 TO THE POWER -5
 SUBTRACT 2 TO THE POWER -7
 SUBTRACT 2 TO THE POWER -9
 SUBTRACT 2 TO THE POWER -12
 C = -1.7763690
 SIGN DIGIT BREAKDOWN OF 1.7763690
 ADD 2 TO THE POWER 1
 SUBTRACT 2 TO THE POWER -2
 ADD 2 TO THE POWER -5
 SUBTRACT 2 TO THE POWER -8
 SUBTRACT 2 TO THE POWER -10
 D = 0.9543562
 SIGN DIGIT BREAKDOWN OF 0.9543562
 ADD 2 TO THE POWER 0
 SUBTRACT 2 TO THE POWER -5
 SUBTRACT 2 TO THE POWER -6
 ADD 2 TO THE POWER -10
 ADD 2 TO THE POWER -12

LOWPASS FILTER
 BREAK FREQUENCY = 1.0000
 SAMPLE FREQUENCY = 200.0000
 ALPHA = -1.9555860
 SIGN DIGIT BREAKDOWN OF 1.9555860
 ADD 2 TO THE POWER 1
 SUBTRACT 2 TO THE POWER -5
 SUBTRACT 2 TO THE POWER -6
 ADD 2 TO THE POWER -9
 ADD 2 TO THE POWER -11
 BETA = 0.9565516
 SIGN DIGIT BREAKDOWN OF 0.9565516
 ADD 2 TO THE POWER 0

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SUBTRACT 2 TO THE POWER -5
SUBTRACT 2 TO THE POWER -6
ADD 2 TO THE POWER -8
SUBTRACT 2 TO THE POWER -11
K1=2** -7
K2=2** -5
K3=2** -10
LARGEST OBSERVED OUTPUT = 0.6799301
BANDPASS FILTER
LOWER CUTOFF FREQUENCY = 2.00000
UPPER CUTOFF FREQUENCY = 10.00000
SAMPLING FREQUENCY = 50.0
A = -1.6701810
SIGN DIGIT BREAKDOWN OF 1.6701810
ADD 2 TO THE POWER 1
SUBTRACT 2 TO THE POWER -2
SUBTRACT 2 TO THE POWER -4
SUBTRACT 2 TO THE POWER -6
SUBTRACT 2 TO THE POWER -9
ADD 2 TO THE POWER -12
B = 0.7390358
SIGN DIGIT BREAKDOWN OF 0.7390358
ADD 2 TO THE POWER -1
ADD 2 TO THE POWER -2
SUBTRACT 2 TO THE POWER -7
SUBTRACT 2 TO THE POWER -8
ADD 2 TO THE POWER -10
SUBTRACT 2 TO THE POWER -12
C = -0.5517535
SIGN DIGIT BREAKDOWN OF 0.5517535
ADD 2 TO THE POWER -1
ADD 2 TO THE POWER -4
SUBTRACT 2 TO THE POWER -7
SUBTRACT 2 TO THE POWER -8
ADD 2 TO THE POWER -10
D = 0.3414242
SIGN DIGIT BREAKDOWN OF 0.3414242
ADD 2 TO THE POWER -2
ADD 2 TO THE POWER -4
ADD 2 TO THE POWER -5
SUBTRACT 2 TO THE POWER -9
SUBTRACT 2 TO THE POWER -11
ADD 2 TO THE POWER -13

LOWPASS FILTER
BREAK FREQUENCY = 0.2000
SAMPLE FREQUENCY = 50.0000
ALPHA = -1.9644670
SIGN DIGIT BREAKDOWN OF 1.9644670
ADD 2 TO THE POWER 1
SUBTRACT 2 TO THE POWER -5
SUBTRACT 2 TO THE POWER -8
SUBTRACT 2 TO THE POWER -11
ADD 2 TO THE POWER -13
BETA = 0.9650878
SIGN DIGIT BREAKDOWN OF 0.9650878
ADD 2 TO THE POWER 0
SUBTRACT 2 TO THE POWER -5
SUBTRACT 2 TO THE POWER -8
ADD 2 TO THE POWER -12
K1=2** -3
K2=2** 0
K3=2** -12
LARGEST OBSERVED OUTPUT = 0.9086537

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*****
* PROGRAM OPERATING IN 2920 NUMBER 1- EEG TAKEN AS INPUT AT*
* A RATE OF 6.5 KHZ, WITH BANDPASS/FW RECT/LOWPASS FILTERS *
* OPERATING AT 200, 50, & 12.5 HZ. THIS 2920 ALSO PRODUCES *
* THE SHIFT CONTROL PULSE WHICH PERMITS THE SHIFT IN OF THE*
* SERIAL DATA. THIS SHIFT PULSE OCCURS ONCE A SECOND,*
* PERMITTING A BURST OF SHIFTS CONTROLLED BY THE EOP PULSE.*
* THE 8 MOST SIGNIFICANT BITS OF A VALUE ARE SHIFTED.*
*
*****
LDA DAR KP2 ; LOAD DAR WITH FS/4 FOR DIGITAL INPUT
LDA WORK1 F1D1 L01 IN1 ; OF CONTROL SIGNAL TO SYNCH TIMERS
SUB WORK1 F1D1 R05 IN1 ; ALSO, 0.25 TO 2.0 HZ BANDPASS 'A' CALC-
SUB WORK1 F1D1 R07 IN1 ; ULATION. THE 0.25 TO 2.0 HZ FILTER IS
SUB WORK1 F1D1 R09 IN1 ; OPERATING AT 50 HZ.
SUB WORK1 F1D1 R11 IN1 ;
LDA WORK2 F1D2 R00 IN1 ; 0.25 TO 2.0 HZ BANDPASS 'B' CALCULATION
SUB WORK2 F1D2 R05 IN1 ;
SUB WORK2 F1D2 R07 IN1 ;
SUB WORK2 F1D2 R10 ; SAMPLE DONE-ALLOW TO SETTLE
SUB WORK2 F1D2 R12 CVT3 ; CNVRT BIT 3 TO DETERMINE IF 1 OR 0
LDA WORK3 F1D3 L01 ; 0.25 TO 2.0 HZ BANDPASS 'C' CALCULATION
SUB WORK3 F1D3 R02 ;
LDA FIFTY KP4 CND3 ; RESET 50 HZ TIMER IF CNTL LINE = '1'
LDA ONE KP3 CND3 ; RESET 1 HZ TIMER IF CNTL LINE='1'
SUB DAR DAR ; CLEAR DAR FOR EEG1 A/D CONVERSION
SUB WORK3 F1D3 R05 IN0 ; SAMPLE EEG1
SUB WORK3 F1D3 R10 IN0 ;
LDA WORK4 F1D4 R00 IN0 ; 0.25 TO 2.0 HZ BANDPASS 'D' CALCULATION
SUB WORK4 F1D4 R02 IN0 ;
ADD WORK4 F1D4 R06 IN0 ;
SUB WORK4 F1D4 R09 IN0 ;
SUB WORK4 F1D4 R13 IN0 ;
SUB WORK1 WORK2 IN0 ; 0.25 TO 2.0 HZ BANDPASS CALCULATION
ADD WORK1 EEG R09 ; ADD IN EEG SAMPLE MULT BY K1
LDA WORK2 WORK1 R00 CND6 ; BP CALCULATION, START A/D CONVERSION
ADD DAR KM2 ; A/D FIXUP - 2920 PROBLEM
SUB WORK2 F1D2 ; BP CALCULATION
SUB WORK3 WORK4 ; BP CALCULATION
ADD WORK3 WORK2 R01 CVT7 ; ADD RESULT OF FIRST SECTION MULT BY K2
LDA WORK4 WORK3 ; BP CALCULATION
SUB WORK4 F1D4 ; BP CALCULATION
LDA WORK2 F2D1 L01 CVT6 ; LOWPASS FILTER FOR 0.25-2.0 HZ BP 'ALPHA'
SUB WORK2 F2D1 R06 ; CALCULATION
SUB WORK2 F2D1 R09 ;
SUB WORK2 F2D1 R12 CVT5 ;
NOP ;
ABA WORK2 WORK4 R11 ; ADD IN RECT BP OUTPUT MULT BY K3
LDA WORK4 F2D2 R00 CVT4 ; 'BETA' CALCULATION
SUB WORK4 F2D2 R06 ;
SUB WORK4 F2D2 R09 ;
SUB WORK2 WORK4 ;
CVT3 ; LP CALCULATION
CND4 ; A/D FIXUP - 2920 PROBLEM
LDA WORK4 WORK2 ; LP CALCULATION
ADD WORK4 F2D1 L01 CVT2 ; LP CALCULATION
CND4 ; A/D FIXUP - 2920 PROBLEM
ADD WORK4 F2D2 ; LP CALCULATION
LDA CNTL KM7 CVT1 ; SET DEFAULT FOR DIGITAL OUTPUT->0
CND4 ; A/D FIXUP - 2920 PROBLEM
SUB FIFTY KP1 R05 ; DECREMENT 50 HZ TIMER
SUB TWOHUND KP1 R05 CVT0 ; DECREMENT 200 HZ TIMER
SUB TWELVE KP1 R07 ; DECREMENT 12.5 HZ TIMER
LDA EEG DAR ; SAVE THE A/D CONVERTED EEG1 SAMPLE
LDA DAR FIFTY ; LOAD 50 HZ TIMER FOR CONDITIONAL DELAYS
LDA F1D2 F1D1 CND5 ; CONDITIONAL DELAY
LDA F1D1 WORK1 CND5 ; CONDITIONAL DELAY
LDA F1D4 F1D3 CND5 ; CONDITIONAL DELAY
LDA F1D3 WORK3 CND5 ; CONDITIONAL DELAY
ADD ONE KM1 R04 CND5 ; DECREMENT 1 HZ TIMER
LDA DAR TWELVE ; LOAD 12.5 HZ TIMER FOR CONDITIONAL DELAYS
LDA F2D2 F2D1 CND5 ; CONDITIONAL DELAY
LDA F2D1 WORK2 CND5 ; CONDITIONAL DELAY
LDA F2OUT WORK4 CND5 ; CONDITIONAL DELAY
LDA TWELVE KP4 CND5 ; RESET 12.5 HZ TIMER
LDA DAR ONE ; LOAD 1 HZ TIMER FOR SHIFT CONTROL
LDA EIGHT KM8 CND5 ; RESET THE COUNT 8 TIMER FOR SHIFTING
LDA TEMP2 F2OUT CND5 ; CATCH FILTER OUTPUTS FOR (DESTRUCTIVE)
LDA TEMP6 F6OUT CND5 ; SHIFT OUT OF VALUES (50 HZ IN SYNCH)
LDA ONE KP3 CND5 ; RESET 1 HZ TIMER
LDA DAR EIGHT ; LOAD COUNT 8 TIMER TO SET CONTROL OUTPUT
ADD EIGHT KP1 CND5 ; 'DECREMENT' COUNT 8 TIMER
LDA CNTL KP7 CND5 ; SET CONTROL OUTPUT HIGH IF IN COUNTDOWN
LDA DAR CNTL ; LOAD CONTROL IN DAR FOR OUTPUT
LDA WORK1 F3D1 L01 ; 2.0 TO 7.0 HZ BANDPASS FILTER 'A' CALC
SUB WORK1 F3D1 R02 ;
SUB WORK1 F3D1 R05 ;
SUB WORK1 F3D1 R09 OUT7 ; SHIFT CONTROL OUTPUT
ADD WORK1 F3D1 R12 OUT7 ; SHIFT CONTROL OUTPUT
LDA CNTL KM7 ; SET DEFAULT DIGITAL OUTPUT TO '0'
LDA DAR TEMP2 ; LOAD CAPTURED OUTPUT FOR SERIAL OUTPUT
LDA CNTL KP7 CND1 ; SEND OUT BIT 1 (LEAST SIG OF UPPER 8 BITS)
LDA TEMP2 DAR R01 ; SAVE BACK SHIFTED VALUE FOR NEXT SHIFT
LDA DAR CNTL ; LOAD DAR WITH CONTROL VALUE FOR OUTPUT
LDA WORK2 F3D2 R00 ; BANDPASS FILTER 'B' CALCULATION
ADD WORK3 F3D3 R04 OUT6 ; AND BANDPASS FILTER 'C' CALCULATION
ADD WORK3 F3D3 R06 ;
ADD WORK3 F3D3 R09 ;
ADD WORK3 F3D3 R10 ;
SUB WORK3 F3D3 R13 ;

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LDA WORK4 F3D4 R01 ; BANDPASS FILTER 'D' CALCULATION
ADD WORK4 F3D4 R05 ;
SUB WORK4 F3D4 R09 ;
SUB WORK4 F3D4 R10 ; 2 TO 7 HZ AMP ESTIMATION OUTPUT ****
SUB WORK1 WORK2 ; BP CALCULATION
LDA CNTL KM7 ; SET DEFAULT DIGITAL OUTPUT TO '0'
LDA DAR TEMP6 ; LOAD CAPTURED OUTPUT FOR SERIAL OUTPUT
LDA CNTL KP7 CND1; SEND OUT BIT 1 (LEAST SIG OF UPPER 8 BITS)
LDA TEMP6 DAR R01 ; SAVE BACK SHIFTED VALUE FOR NEXT SHIFT
LDA DAR CNTL ; LOAD DAR WITH CONTROL VALUE FOR OUTPUT
ADD WORK1 EEG R04 ; ADD IN EEG SAMPLE MULT BY K1
SUB WORK2 F3D2 ; BP CALCULATION
LDA WORK2 WORK1 ; BP CALCULATION
SUB WORK3 WORK4 ; 12 TO 14 HZ AMP ESTIMATION OUTPUT ****
ADD WORK3 WORK2 R00 OUT4; ADD FIRST SECTION OUTPUT MULT BY K2
LDA WORK4 WORK3 ; BP CALCULATION
SUB WORK4 F3D4 ; BP CALCULATION
LDA WORK2 F4D1 L01 ; 2 - 7 HZ BP/FW RECT/ LOWPASS 'ALPHA' CALC
SUB WORK2 F4D1 R05 ;
SUB WORK2 F4D1 R08 ;
SUB WORK2 F4D1 R11 ;
ADD WORK2 F4D1 R13 ;
ABA WORK2 WORK4 R12 ; ADD RECT BP OUTPUT MULT BY K3
LDA WORK4 F4D2 R00 ; LOWPASS FILTER 'BETA' CALCULATION
SUB WORK4 F4D2 R05 ;
SUB WORK4 F4D2 R08 ;
ADD WORK4 F4D2 R12 ;
SUB WORK2 WORK4 ; LP CALCULATION
LDA WORK4 WORK2 ; LP CALCULATION
ADD WORK4 F4D1 L01 ; LP CALCULATION
ADD WORK4 F4D2 R00 ; LP CALCULATION
LDA DAR FIFTY ; LOAD 50 HZ TIMER FOR CONDITIONAL DELAYS
LDA F3D2 F3D1 CND5; CONDITIONAL DELAY
LDA F3D1 WORK1 CND5; CONDITIONAL DELAY
LDA F3D4 F3D3 CND5; CONDITIONAL DELAY
LDA F3D3 WORK3 CND5; CONDITIONAL DELAY
LDA F4D2 F4D1 CND5; CONDITIONAL DELAY
LDA F4D1 WORK2 CND5; CONDITIONAL DELAY
LDA F4OUT WORK4 CND5; CONDITIONAL DELAY
LDA FIFTY KP4 CND5; RESET 50 HZ TIMER
LDA CNTL KM7 ; SET DEFAULT DIGITAL OUTPUT TO '0'
LDA DAR F4OUT ; LOAD CAPTURED OUTPUT FOR SERIAL OUTPUT
LDA CNTL KP7 CND1; SEND OUT BIT 1 (LEAST SIG OF UPPER 8 BITS)
LDA F4OUT DAR R01 ; SAVE BACK SHIFTED VALUE FOR NEXT SHIFT
LDA DAR CNTL ; LOAD DAR WITH CONTROL VALUE FOR OUTPUT
LDA WORK1 F5D1 ; 12 TO 14 HZ BANDPASS FILTER 'A' CALC
SUB WORK1 F5D1 R01 ;
SUB WORK1 F5D1 R03 ;
SUB WORK1 F5D1 R04 ;
ADD WORK1 F5D1 R09 OUT5;
ADD WORK1 F5D1 R11 OUT5;
LDA WORK2 F5D2 R00 ; BANDPASS FILTER 'B' CALCULATION
SUB WORK2 F5D2 R05 ;
SUB WORK2 F5D2 R07 ;
SUB WORK2 F5D2 R09 ;
SUB WORK2 F5D2 R12 ;
LDA WORK3 F5D3 L01 ; BANDPASS FILTER 'C' CALCULATION
SUB WORK3 F5D3 R02 ;
ADD WORK3 F5D3 R05 ;
SUB WORK3 F5D3 R08 ;
SUB WORK3 F5D3 R10 ;
LDA WORK4 F5D4 R00 ; BANDPASS FILTER 'D' CALCULATION
SUB WORK4 F5D4 R05 ;
SUB WORK4 F5D4 R06 ;
ADD WORK4 F5D4 R10 ;
ADD WORK4 F5D4 R12 ;
SUB WORK1 WORK2 ; BP CALCULATION
ADD WORK1 EEG R07 ; ADD IN EEG1 SAMPLE MULT BY K1
LDA WORK2 WORK1 ; BP CALCULATION
SUB WORK2 F5D2 ; BP CALCULATION
SUB WORK3 WORK4 ; BP CALCULATION
ADD WORK3 WORK2 R05 ; ADD FIRST SECTION OUTPUT MULT BY K2
LDA WORK4 WORK3 ; BP CALCULATION
SUB WORK4 F5D4 ; BP CALCULATION
LDA WORK2 F6D1 L01 ; 12 - 14 HZ BP/RECT/LOWPASS 'ALPHA' CALC
SUB WORK2 F6D1 R05 ;
SUB WORK2 F6D1 R06 ;
ADD WORK2 F6D1 R09 ;
ADD WORK2 F6D1 R11 ;
ABA WORK2 WORK4 R10 ; ADD IN THE FW RECTIFIED MULT BY K3
LDA WORK4 F6D2 R00 ; 'BETA' CALCULATION
SUB WORK4 F6D2 R05 ;
SUB WORK4 F6D2 R06 ;
ADD WORK4 F6D2 R08 ;
SUB WORK4 F6D2 R11 ;
SUB WORK2 WORK4 ; LP CALCULATION
LDA WORK4 WORK2 ; LP CALCULATION
ADD WORK4 F6D1 L01 ; LP CALCULATION
ADD WORK4 F6D2 R00 ; LP CALCULATION
LDA DAR TWOHUND ; LOAD 200 HZ TIMER FOR CONDITIONAL DELAYS
LDA F5D2 F5D1 CND5; CONDITIONAL DELAY
LDA F5D1 WORK1 CND5; CONDITIONAL DELAY
LDA F5D4 F5D3 CND5; CONDITIONAL DELAY
LDA F5D3 WORK3 CND5; CONDITIONAL DELAY
LDA F6D2 F6D1 CND5; CONDITIONAL DELAY
LDA F6D1 WORK2 CND5; CONDITIONAL DELAY
LDA F6OUT WORK4 CND5; CONDITIONAL DELAY
LDA TWOHUND KP1 EOP; SIGNAL END OF PROGRAM (PREFETCHED 4 INST)
CND5; RESET 200 HZ TIMER
NOP ;
NOP ;

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*****
* PROGRAM OPERATING IN 2920 NUMBER 2- EEG TAKEN AS INPUT AT*
* A RATE OF 6.5 KHZ, WITH BANDPASS/FW RECT/LOWPASS FILTERS *
* OPERATING AT 200, 50, & 12.5 HZ. THIS 2920 USES THE *
* SHIFT CONTROL PULSE WHICH IS PROVIDED BY 2920 NUMBER 1. *
* THIS SHIFT PULSE OCCURS ONCE A SECOND, PERMITTING A BURST*
* OF SHIFTS CONTROLLED BY THE EOP PULSE. THE 8 MOST SIGNIF *
* BITS OF A VALUE ARE SHIFTED. THE 2920S ARE ALL SYNCED *
* WITH A CONTROL INPUT AND WITH THEIR EOPS TIED TOGETHER. *
*****
LDA DAR KP2 ; LOAD DAR WITH FS/4 FOR DIGITAL INPUT
LDA WORK1 F1D1 L01 IN1 ; OF CONTROL SIGNAL TO SYNCH TIMERS
SUB WORK1 F1D1 R05 IN1 ; ALSO, 0.25 TO 2.0 HZ BANDPASS 'A' CALC-
SUB WORK1 F1D1 R07 IN1 ; ULATION. THE 0.25 TO 2.0 HZ FILTER IS
SUB WORK1 F1D1 R09 IN1 ; OPERATING AT 50 HZ.
SUB WORK1 F1D1 R11 IN1 ;
LDA WORK2 F1D2 R00 IN1 ; 0.25 TO 2.0 HZ BANDPASS 'B' CALCULATION
SUB WORK2 F1D2 R05 IN1 ;
SUB WORK2 F1D2 R07 IN1 ;
SUB WORK2 F1D2 R10 ;
SUB WORK2 F1D2 R12 CVT3 ; SAMPLE DONE-ALLOW TO SETTLE
LDA WORK3 F1D3 R01 IN1 ; CNVRT BIT 3 TO DETERMINE IF 1 OR 0
SUB WORK3 F1D3 R02 ; 0.25 TO 2.0 HZ BANDPASS 'C' CALCULATION
LDA FIFTY KP4 CND3 ; RESET 50 HZ TIMER IF CNTL LINE='1'
LDA ONE KP3 CND3 ; RESET 1 HZ TIMER IF CNTL LINE='1'
SUB DAR DAR ; CLEAR DAR FOR EEG1 A/D CONVERSION
SUB WORK3 F1D3 R05 IN0 ; SAMPLE EEG1
SUB WORK3 F1D3 R10 IN0 ;
LDA WORK4 F1D4 R00 IN0 ; 0.25 TO 2.0 HZ BANDPASS 'D' CALCULATION
SUB WORK4 F1D4 R02 IN0 ;
ADD WORK4 F1D4 R06 IN0 ;
SUB WORK4 F1D4 R09 IN0 ;
SUB WORK4 F1D4 R13 IN0 ;
SUB WORK1 WORK2 R09 IN0 ; 0.25 TO 2.0 HZ BANDPASS CALCULATION
ADD WORK1 EEG R09 ; ADD IN EEG SAMPLE MULT BY K1
LDA WORK2 WORK1 CVT5 ; BP CALCULATION, START A/D CONVERSION
ADD DAR KM2 R00 CND6 ; A/D FIXUP - 2920 PROBLEM
SUB WORK2 F1D2 ; BP CALCULATION
SUB WORK3 WORK4 R01 CVT7 ; BP CALCULATION
ADD WORK3 WORK2 ; ADD RESULT OF FIRST SECTION MULT BY K2
LDA WORK4 WORK3 ; BP CALCULATION
SUB WORK4 F1D4 ; BP CALCULATION
LDA WORK2 F2D1 L01 CVT6 ; LOWPASS FILTER FOR 0.25-2.0 HZ BP 'ALPHA'
SUB WORK2 F2D1 R06 ; CALCULATION
SUB WORK2 F2D1 R09 ;
SUB WORK2 F2D1 R12 NOP ;
ABA WORK2 WORK4 R11 CVT5 ; ADD IN RECT BP OUTPUT MULT BY K3
LDA WORK4 F2D2 R00 CVT4 ; 'BETA' CALCULATION
SUB WORK4 F2D2 R06 ;
SUB WORK4 F2D2 R09 ;
SUB WORK2 WORK4 CVT3 ; LP CALCULATION
LDA WORK4 WORK2 CND4 ; A/D FIXUP - 2920 PROBLEM
ADD WORK4 F2D1 L01 CVT2 ; LP CALCULATION
ADD WORK4 F2D2 CND4 ; LP CALCULATION
LDA CNTL KM7 CVT1 ; A/D FIXUP - 2920 PROBLEM
SUB FIFTY KP1 R05 CVT4 ; SET DEFAULT FOR DIGITAL OUTPUT->0
SUB TWOHUND KP1 R05 CVT0 ; A/D FIXUP - 2920 PROBLEM
SUB TWELVE KP1 R07 ; DECREMENT 50 HZ TIMER
LDA EEG DAR ; DECREMENT 200 HZ TIMER
LDA DAR FIFTY ; DECREMENT 12.5 HZ TIMER
LDA F1D2 F1D1 CND5 ; SAVE THE A/D CONVERTED EEG1 SAMPLE
LDA F1D1 WORK1 CND5 ; LOAD 50 HZ TIMER FOR CONDITIONAL DELAYS
LDA F1D4 F1D3 CND5 ; CONDITIONAL DELAY
LDA F1D3 WORK3 CND5 ; CONDITIONAL DELAY
ADD ONE KM1 R04 CND5 ; CONDITIONAL DELAY
LDA DAR TWELVE ; DECREMENT 1 HZ TIMER
LDA F2D2 F2D1 CND5 ; LOAD 12.5 HZ TIMER FOR CONDITIONAL DELAYS
LDA F2D1 WORK2 CND5 ; CONDITIONAL DELAY
LDA F2OUT WORK4 CND5 ; CONDITIONAL DELAY
LDA TWELVE KP4 CND5 ; CONDITIONAL DELAY
LDA DAR ONE CND5 ; RESET 12.5 HZ TIMER
LDA EIGHT KM8 CND5 ; LOAD 1 HZ TIMER FOR SHIFT CONTROL
LDA TEMP2 F2OUT CND5 ; RESET THE COUNT 8 TIMER FOR SHIFTING
LDA TEMP6 F6OUT CND5 ; CATCH FILTER OUTPUTS FOR (DESTRUCTIVE)
LDA ONE KP3 CND5 ; SHIFT OUT OF VALUES (50 HZ IN SYNC)
LDA DAR EIGHT CND5 ; RESET 1 HZ TIMER
ADD EIGHT KP1 CND5 ; LOAD COUNT 8 TIMER TO SET CONTROL OUTPUT
LDA WORK1 F3D1 L01 ; 'DECREMENT' COUNT 8 TIMER
SUB WORK1 F3D1 R02 ; 2.0 TO 7.0 HZ BANDPASS FILTER 'A' CALC
SUB WORK1 F3D1 R05 ;
SUB WORK1 F3D1 R09 ;
ADD WORK1 F3D1 R12 ;
LDA DAR TEMP2 ;
LDA CNTL KP7 CND1 ; LOAD CAPTURED OUTPUT FOR SERIAL OUTPUT
LDA TEMP2 DAR R01 ; SEND OUT BIT 1 (LEAST SIG OF UPPER 8 BITS)
LDA DAR CNTL ; SAVE BACK SHIFTED VALUE FOR NEXT SHIFT
LDA WORK2 F3D2 R00 ; LOAD DAR WITH CONTROL VALUE FOR OUTPUT
SUB WORK2 F3D2 R02 ; BANDPASS FILTER 'B' CALCULATION
ADD WORK2 F3D2 R05 ;
LDA WORK3 F3D3 R00 OUT6 ; .25 TO 2 HZ AMP ESTIMATION OUTPUT ****
ADD WORK3 F3D3 R04 OUT6 ; AND BANDPASS FILTER 'C' CALCULATION
ADD WORK3 F3D3 R06 ;
ADD WORK3 F3D3 R09 ;
ADD WORK3 F3D3 R10 ;

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SUB WORK3 F3D3 R13 ;
LDA WORK4 F3D4 R01 ; BANDPASS FILTER 'D' CALCULATION
ADD WORK4 F3D4 R05 ;
SUB WORK4 F3D4 R09 ;
SUB WORK4 F3D4 R10 ; 2 TO 7 HZ AMP ESTIMATION OUTPUT ****
SUB WORK1 WORK2 ; BP CALCULATION
LDA CNTL KM7 ; SET DEFAULT DIGITAL OUTPUT TO '0'
LDA DAR TEMP6 ; LOAD CAPTURED OUTPUT FOR SERIAL OUTPUT
LDA CNTL KP7 CND1; SEND OUT BIT 1 (LEAST SIG OF UPPER 8 BITS)
LDA TEMP6 DAR R01; SAVE BACK SHIFTED VALUE FOR NEXT SHIFT
LDA DAR CNTL ; LOAD DAR WITH CONTROL VALUE FOR OUTPUT
ADD WORK1 EEG R04; ADD IN EEG SAMPLE MULT BY K1
LDA WORK2 WORK1 ; BP CALCULATION
SUB WORK2 F3D2 ; BP CALCULATION
SUB WORK3 WORK4 ; 12 TO 14 HZ AMP ESTIMATION OUTPUT ****
ADD WORK3 WORK2 R00; OUT4; ADD FIRST SECTION OUTPUT MULT BY K2
LDA WORK4 WORK3 ; BP CALCULATION
SUB WORK4 F3D4 ; BP CALCULATION
LDA WORK2 F4D1 L01; 2 - 7 HZ BP/FW RECT/ LOWPASS 'ALPHA' CALC
SUB WORK2 F4D1 R05 ;
SUB WORK2 F4D1 R08 ;
SUB WORK2 F4D1 R11 ;
ADD WORK2 F4D1 R13 ;
ABA WORK2 WORK4 R12; ADD RECT BP OUTPUT MULT BY K3
LDA WORK4 F4D2 R00; LOWPASS FILTER 'BETA' CALCULATION
SUB WORK4 F4D2 R05 ;
SUB WORK4 F4D2 R08 ;
ADD WORK4 F4D2 R12 ;
SUB WORK2 WORK4 ; LP CALCULATION
LDA WORK4 WORK2 ; LP CALCULATION
ADD WORK4 F4D1 L01; LP CALCULATION
ADD WORK4 F4D2 R00; LP CALCULATION
LDA DAR FIFTY ; LOAD 50 HZ TIMER FOR CONDITIONAL DELAYS
LDA F3D2 F3D1 CND5; CONDITIONAL DELAY
LDA F3D1 WORK1 CND5; CONDITIONAL DELAY
LDA F3D4 F3D3 CND5; CONDITIONAL DELAY
LDA F3D3 WORK3 CND5; CONDITIONAL DELAY
LDA F4D2 F4D1 CND5; CONDITIONAL DELAY
LDA F4D1 WORK2 CND5; CONDITIONAL DELAY
LDA F4OUT WORK4 CND5; CONDITIONAL DELAY
LDA FIFTY KP4 CND5; RESET 50 HZ TIMER
LDA CNTL KM7 ; SET DEFAULT DIGITAL OUTPUT TO '0'
LDA DAR F4OUT ; LOAD CAPTURED OUTPUT FOR SERIAL OUTPUT
LDA CNTL KP7 CND1; SEND OUT BIT 1 (LEAST SIG OF UPPER 8 BITS)
LDA F4OUT DAR R01; SAVE BACK SHIFTED VALUE FOR NEXT SHIFT
LDA DAR CNTL ; LOAD DAR WITH CONTROL VALUE FOR OUTPUT
LDA WORK1 F5D1 L01; 12 TO 14 HZ BANDPASS FILTER 'A' CALC
SUB WORK1 F5D1 R03 ;
SUB WORK1 F5D1 R04 ;
ADD WORK1 F5D1 R09; OUT5;
ADD WORK1 F5D1 R11; OUT5;
LDA WORK2 F5D2 R00; BANDPASS FILTER 'B' CALCULATION
SUB WORK2 F5D2 R05 ;
SUB WORK2 F5D2 R07 ;
SUB WORK2 F5D2 R09 ;
SUB WORK2 F5D2 R12 ;
LDA WORK3 F5D3 L01; BANDPASS FILTER 'C' CALCULATION
SUB WORK3 F5D3 R02 ;
ADD WORK3 F5D3 R05 ;
SUB WORK3 F5D3 R08 ;
SUB WORK3 F5D3 R10 ;
LDA WORK4 F5D4 R00; BANDPASS FILTER 'D' CALCULATION
SUB WORK4 F5D4 R05 ;
SUB WORK4 F5D4 R06 ;
ADD WORK4 F5D4 R10 ;
ADD WORK4 F5D4 R12 ;
SUB WORK1 WORK2 ; BP CALCULATION
ADD WORK1 EEG R07; ADD IN EEG1 SAMPLE MULT BY K1
LDA WORK2 WORK1 ; BP CALCULATION
SUB WORK2 F5D2 ; BP CALCULATION
SUB WORK3 WORK4 ; BP CALCULATION
ADD WORK3 WORK2 R05; ADD FIRST SECTION OUTPUT MULT BY K2
LDA WORK4 WORK3 ; BP CALCULATION
SUB WORK4 F5D4 ; BP CALCULATION
LDA WORK2 F6D1 L01; 12 - 14 HZ BP/RECT/LOWPASS 'ALPHA' CALC
SUB WORK2 F6D1 R05 ;
SUB WORK2 F6D1 R06 ;
ADD WORK2 F6D1 R09 ;
ADD WORK2 F6D1 R11 ;
ABA WORK2 WORK4 R10; ADD IN THE FW RECTIFIED MULT BY K3
LDA WORK4 F6D2 R00; 'BETA' CALCULATION
SUB WORK4 F6D2 R05 ;
SUB WORK4 F6D2 R06 ;
ADD WORK4 F6D2 R08 ;
SUB WORK4 F6D2 R11 ;
SUB WORK2 WORK4 ; LP CALCULATION
LDA WORK4 WORK2 ; LP CALCULATION
ADD WORK4 F6D1 L01; LP CALCULATION
ADD WORK4 F6D2 R00; LP CALCULATION
LDA DAR TWOHUND ; LOAD 200 HZ TIMER FOR CONDITIONAL DELAYS
LDA F5D2 F5D1 CND5; CONDITIONAL DELAY
LDA F5D1 WORK1 CND5; CONDITIONAL DELAY
LDA F5D4 F5D3 CND5; CONDITIONAL DELAY
LDA F5D3 WORK3 CND5; CONDITIONAL DELAY
LDA F6D2 F6D1 CND5; CONDITIONAL DELAY
LDA F6D1 WORK2 CND5; CONDITIONAL DELAY
LDA F6OUT WORK4 CND5; CONDITIONAL DELAY
LDA TWOHUND KP1 CND5; RESET 200 HZ TIMER
NOP ; NOPS FOR REMAINDER OF
NOP ; FULL 192 PROGRAM STEPS

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*****
* PROGRAM OPERATING IN 2920 NUMBER 3- EOG TAKEN AS INPUT AT*
* A RATE OF 6.5 KHZ, WITH BANDPASS/FW RECT/LOWPASS FILTER *
* OPERATING AT 50 HZ. THIS 2920 USES THE SHIFT CONTROL *
* PULSE WHICH IS PROVIDED BY 2920 NUMBER 1. *
* THIS SHIFT PULSE OCCURS ONCE A SECOND, PERMITTING A BURST*
* OF SHIFTS CONTROLLED BY THE EOP PULSE. THE 8 MOST SIGNIF *
* BITS OF A VALUE ARE SHIFTED. THE 2920S ARE ALL SYNCED *
* WITH A CONTROL INPUT AND WITH THEIR EOPS TIED TOGETHER. *
*****
LDA DAR KP2 ; LOAD DAR WITH FS/4 FOR DIGITAL INPUT
LDA WORK1 F1D1 L01 IN1 ; OF CONTROL SIGNAL TO SYNCH TIMERS
SUB WORK1 F1D1 R02 IN1 ; ALSO, 2.0 TO 10 HZ BANDPASS 'A' CALC-
SUB WORK1 F1D1 R04 IN1 ; ULATION. THE 2.0 TO 10 HZ FILTER IS
SUB WORK1 F1D1 R06 IN1 ; OPERATING AT 50 HZ.
SUB WORK1 F1D1 R09 IN1 ;
ADD WORK1 F1D1 R12 IN1 ;
LDA WORK2 F1D2 R01 IN1 ; 2.0 TO 10 HZ BANDPASS 'B' CALCULATION
ADD WORK2 F1D2 R02 IN1 ;
SUB WORK2 F1D2 R07 ; SAMPLE DONE-ALLOW TO SETTLE
SUB WORK2 F1D2 R08 CVT3; CNVRT BIT 3 TO DETERMINE IF 1 OR 0
ADD WORK2 F1D2 R10 ;
SUB WORK2 F1D2 R12 ;
LDA FIFTY KP4 ; CND3; RESET 50 HZ TIMER IF CNTL LINE = '1'
LDA ONE KP3 ; CND3; RESET 1 HZ TIMER IF CNTL LINE='1'
SUB DAR DAR ; CLEAR DAR FOR EOG A/D CONVERSION
LDA WORK3 F1D3 R01 IN0 ; SAMPLE EOG, BANDPASS 'C' CALCULATION
ADD WORK3 F1D3 R04 IN0 ;
SUB WORK3 F1D3 R07 IN0 ;
SUB WORK3 F1D3 R08 IN0 ;
ADD WORK3 F1D3 R10 IN0 ;
LDA WORK4 F1D4 R02 IN0 ; BANDPASS 'D' CALCULATION
ADD WORK4 F1D4 R04 IN0 ;
ADD WORK4 F1D4 R05 IN0 ;
SUB WORK4 F1D4 R09 ;
SUB WORK4 F1D4 R11 CVT5; START A/D CONVERSION
ADD DAR KM2 R00 CND6; A/D FIXUP - 2920 PROBLEM
ADD WORK4 F1D4 R13 ;
SUB WORK1 WORK2 ; BP CALCULATION
ADD WORK1 EOG R03 CVT7; ADD IN SAMPLE MULT BY K1
LDA WORK2 WORK1 ; BP CALCULATION
SUB WORK2 F1D2 ; BP CALCULATION
SUB WORK3 WORK4 ; BP CALCULATION
ADD WORK3 WORK2 R00 ; ADD RESULT OF FIRST SECTION MULT BY K2
LDA WORK4 WORK3 ; BP CALCULATION
SUB WORK4 F1D4 ; BP CALCULATION
LDA WORK2 F2D1 L01 ; LOWPASS FILTER 'ALPHA' CALCULATION
SUB WORK2 F2D1 R05 ;
SUB WORK2 F2D1 R08 CVT4;
SUB WORK2 F2D1 R11 ;
ADD WORK2 F2D1 R13 ;
ABS WORK4 WORK4 ; CVT3; FULL WAVE RECTIFY THE BP OUTPUT
; CND4; A/D FIXUP - 2920 PROBLEM
ADD WORK2 WORK4 R12 ; ADD IN SECOND SECTION MULT BY K3
LDA WORK4 F2D2 R00 CVT2; LOWPASS FILTER 'BETA' CALCULATION
; CND4; A/D FIXUP - 2920 PROBLEM
SUB WORK4 F2D2 R05 ;
SUB WORK4 F2D2 R08 CVT1;
; CND4; A/D FIXUP - 2920 PROBLEM
ADD WORK4 F2D2 R12 ;
SUB WORK2 WORK4 ; CVT0; LP CALCULATION
LDA WORK4 WORK2 ; LP CALCULATION
ADD WORK4 F2D1 L01 ; LP CALCULATION
ADD WORK4 F2D2 ; LP CALCULATION
LDA EOG DAR ; SAVE THE A/D CONVERTED EOG SAMPLE
LDA CNTL KM7 ; SET DEFAULT FOR DIGITAL OUTPUT->0
SUB FIFTY KP1 R05 ; DECREMENT 50 HZ TIMER
LDA DAR FIFTY ; LOAD 50 HZ TIMER FOR CONDITIONAL DELAYS
LDA F1D2 F1D1 ; CND5; CONDITIONAL DELAY
LDA F1D1 WORK1 ; CND5; CONDITIONAL DELAY
LDA F1D4 F1D3 ; CND5; CONDITIONAL DELAY
LDA F1D3 WORK3 ; CND5; CONDITIONAL DELAY
ADD ONE KM1 R04 ; CND5; DECREMENT 1 HZ TIMER
LDA F2D2 F2D1 ; CND5; CONDITIONAL DELAY
LDA F2D1 WORK2 ; CND5; CONDITIONAL DELAY
LDA F2OUT WORK4 ; CND5; CONDITIONAL DELAY
LDA FIFTY KP4 ; CND5; RESET THE 50 HZ TIMER
LDA DAR ONE ; LOAD 1 HZ TIMER FOR SHIFT CONTROL
LDA EIGHT KM8 ; CND5; RESET THE COUNT 8 TIMER FOR SHIFTING
LDA ONE KP3 ; CND5; RESET 1 HZ TIMER
LDA DAR EIGHT ; LOAD COUNT 8 TIMER
ADD EIGHT KP1 ; CND5; DECREMENT COUNT 8 TIMER
LDA DAR F2OUT ; LOAD LP OUTPUT FOR SHIFTOUT
LDA CNTL KP7 ; CND1; SET OUT BIT 1(LEAST SIG OF UPPER 8 BITS)
LDA F2OUT DAR R01 ; SAVE BACK SHIFTED VALUE FOR NEXT SHIFT
LDA DAR CNTL ; LOAD DAR WITH CONTROL VALUE FOR OUTPUT
; NOP
; NOP
SUB WORK2 F3D2 R02 ;
ADD WORK2 F3D2 R05 ;
LDA WORK3 F3D3 R00 ; OUT6; .25 TO 2 HZ AMP ESTIMATION OUTPUT ****
; NOP
; OUT6; 2.0 TO 10 HZ AMP ESTIMATION OUTPUT ****
; OUT6;
; NOP ; NOPS FOR REMAINDER OF FULL 192
; NOP ; PROGRAM STEPS

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//REIMER JOB ',,T=15,R=256,I=30,C=0,L=5','REIMER'
/*D800 VPLLOT
// EXEC FORTHCLG,USERLIB='SYS2.VPLOTLIB'
//FORT.SYSIN DD *
C *****
C *
C * PROGRAM TO APPLY A GIVEN EPOCH RULE (ADJUSTABLE LENGTH) AND *
C * PLOT THE RESULTING IMAGE ON THE VERSATEC PLOTTER *
C *
C *****
        INTEGER IBUF(4000),INDEX,INDEX1,N,NOW, LAST
        REAL INCR,EPOCH,AVER1,AVER2,AVER3
        REAL BPM1(3000),PAGE(3000),SECS(3000),TIME(3000),PGS
        REAL DUMMY,LARGE,Y,X,BPL1(3000),BPH1(3000),BPL,BPM,BPH
61      FORMAT(F4.0,4(F5.0))
        BPL=0
        BPM=0
        BPH=0
        INDEX=1
1      READ(5,61,END=10) PAGE(INDEX),SECS(INDEX),BPL1(INDEX),
        * BPM1(INDEX),BPH1(INDEX)
        TIME(INDEX)=PAGE(INDEX)+(SECS(INDEX)/30.0)
        IF(BPL1(INDEX).GT. BPL) BPL=BPL1(INDEX)
        IF(BPM1(INDEX).GT. BPM) BPM=BPM1(INDEX)
        IF(BPH1(INDEX).GT. BPH) BPH=BPH1(INDEX)
        INDEX=INDEX+1
10     GO TO 1
        LAST=INDEX-1
        DO 20 INDEX=1,LAST
            BPL1(INDEX)=BPL1(INDEX)/BPL
            BPM1(INDEX)=BPM1(INDEX)/BPM
            BPH1(INDEX)=BPH1(INDEX)/BPH
20     CONTINUE
        N=0
        AVER1=0
        AVER2=0
        AVER3=0
        INCR=10.0/30.0
        PGS=PAGE(LAST)-PAGE(1)+1
        PGS=0.25*PGS
        CALL SIZE(PGS)
        CALL PLOTS(IBUF,4000)
        EPOCH=TIME(1)+INCR
        DO 6 INDEX=1,LAST
            N=N+1
            AVER1=AVER1+BPL1(INDEX)
            AVER2=AVER2+BPM1(INDEX)
            AVER3=AVER3+BPH1(INDEX)
            IF(TIME(INDEX).LT. EPOCH) GO TO 6
            AVER1=AVER1/N
            AVER2=AVER2/N
            AVER3=AVER3/N
            DO 5 INDEX1=1,N
                NOW=INDEX-INDEX1+1
                BPL1(NOW)=AVER1
                BPM1(NOW)=AVER2
                BPH1(NOW)=AVER3
5            CONTINUE
            N=0
            AVER1=0
            AVER2=0
            AVER3=0
            EPOCH=EPOCH+INCR
6        CONTINUE
        Y=1.0/4.0
        X=4.0
        CALL PLOT(0.0,-PGS,-3)
        CALL PLOT(0.0,0.5,-3)
        TIME(LAST+1)=PAGE(1)
        BPM1(LAST+1)=0.0
        TIME(LAST+2)=X
        BPL1(LAST+2)=Y
        CALL AXIS(0.0,0.0,'0.25 TO 2.0',11,4.0,90.0,0.0,0.0,Y)
        CALL AXIS(0.0,0.0,'TIME(PAGE)',-10,PGS,0.0,PAGE(1),X)
        CALL LINE(TIME,BPL1,LAST,1,0,0)
        CALL PLOT(0.0,4.1,-3)
        BPM1(LAST+1)=0.0
        BPM1(LAST+2)=Y
        CALL AXIS(0.0,0.0,'2.0 TO 7.0',10,4.0,90.0,0.0,0.0,Y)
        CALL LINE(TIME,BPM1,LAST,1,0,0)
        CALL PLOT(0.0,4.1,-3)
        BPH1(LAST+1)=0.0
        BPH1(LAST+2)=0.5
        CALL AXIS(0.0,0.0,'12 TO 14',8,2.0,90.0,0.0,0.0,0.5)
        CALL LINE(TIME,BPH1,LAST,1,0,0)
        CALL PLOT((PGS+4.0),0.0,999)
        STOP
        END
/*
//GO.FT01F001 DD DSN=&FT01F001,UNIT=SYSDA,SPACE=(CYL,(2,2)),
// DISP=(NEW,PASS)
//GO.VWORK DD DSN=&VWORK,UNIT=SYSDA,SPACE=(CYL,(2,2)),
// DISP=(NEW,PASS)
//GO.SYSIN DD DSN=JREIMER.PLOT.DATA,VOL=SER=USER02,
// DCB=(LRECL=80,BLKSIZE=6080),UNIT=SYSDA,DISP=(OLD,KEEP)
// EXEC VPLLOT,COND=(0,NE)

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*****
*
* PROGRAM TO ALLOW THE EXORCISOR TO APPEAR AS AN INTELLEAGENT
* TERMINAL TO THE AMDAHL COMPUTER
*
*****
0010 TERM:
0020 PROC OPTIONS(MAIN)
0030 $ NAM TERM
0040 DCL
0050 1 IOCB,
0060 2 STATUS BIN(1),
0070 2 MODE BIN(1),
0080 2 BUFP BIN(2),
0090 2 BUFS BIN(2),
0100 2 BUFE BIN(2),
0110 2 TYPE CHAR(2),
0120 2 UNIT CHAR(1),
0130 2 NAME CHAR(8),
0140 2 SUF CHAR(2),
0150 2 DUM0 BIN(2),
0160 2 FORMAT BIN(1),
0170 2 DUM1 BIN(3),
0180 2 DUM2 BIN(2),
0190 2 ALLOC BIN(2),
0200 2 SECT5 BIN(2),
0210 2 SECTE BIN(2),
0220 2 SECTP BIN(2)
0230 DCL INPUT CHAR(8)
0240 DCL DUMMY CHAR(1)
0250 DCL INPUT1 CHAR(13)
0260 DCL INDEX1 BIN(1)
0270 DCL PRMPT CHAR(12) INIT('EXORCISOR=>')
0280 DCL EOTCH CHAR(1) INIT($4)
0290 DCL CMD1 CHAR(8) INIT('ONLINE')
0300 DCL CMD2 CHAR(8) INIT('END')
0310 DCL CMD3 CHAR(8) INIT('UPLOAD')
0320 DCL CMD5 CHAR(8) INIT('HELP')
0330 DCL BUFFER CHAR(60)
0340 DCL BUFEND CHAR(1) INIT($D)
0350 DECLARE
0360 1 BUF,
0370 2 BUFF(133) CHAR(1)
0380 DCL LAST BIN(2)
0390 DCL HOLD CHAR(2)
0400 DCL SECT0 CHAR(256)
0410 DCL SECT1 CHAR(256)
0420 DCL SECT2 CHAR(256)
0430 DCL SECT3 CHAR(256)
0440 DCL SECT4 CHAR(256)
0450 DCL SECT5 CHAR(256)
0460 DCL SECT6 CHAR(256)
0470 DCL SECT7 CHAR(256)
0480 DCL SECT8 CHAR(256)
0490 DCL SECT9 CHAR(256)
0500 DCL INDEX BIN(1)
0510 DCL PACK1 BIN(2)
0520 DCL PACK2 BIN(2)
0530 DCL 1 OUT4,
0540 2 XX7 CHAR(19) INIT('ENTER NAME OF FILE'),
0550 2 XX8 CHAR(15) INIT(' TO BE UPLOADED'),
0560 2 XX9 CHAR(1) INIT($D)
0570 DCL FEED CHAR(1) INIT($C)
0580 DCL LINE CHAR(1) INIT($A)
0590 DCL LCU CHAR(1) INIT($75)
0600 DCL CNTT CHAR(1) INIT($14)
0610 DCL CNTQ CHAR(1) INIT($11)
0620 DCL ESCP CHAR(1) INIT($1B)
0630 DCL CNTO CHAR(1) INIT($0F)
0640 DCL CNTN CHAR(1) INIT($DE)
0650 DCL HOLD0 BIN(1)
0660 DCL HOLD1 BIN(2)
0670 DCL HOLD2 CHAR(1)
0680 DCL NUM BIN(1)
0690 DCL ONE CHAR(1)
0700 DCL LNUM BIN(2)
0710 DCL COUNT2 BIN(2) DEF $E03C
0720 DCL LATCH3 BIN(2) DEF $E03E
0730 DCL LATCH2 BIN(2) DEF $E03C
0740 DCL CNTL3 BIN(1) DEF $E038
0750 DCL CNTL2 BIN(1) DEF $E039
0760 DCL CNTL1 BIN(1) DEF $E038
0770 DCL TIME BIN(2)
0780 DCL TERMCT BIN(1) DEF $FCF4
0790 DCL TERMTR BIN(1) DEF $FCF5
0800 DCL CARDCT BIN(1) DEF $E000
0810 DCL CARDTR BIN(1) DEF $E001
0820 DCL STATE BIN(1)
0830 DCL 1 OUT6,
0840 2 XX10 CHAR(18) INIT('SENDING RECORD #: '),
0850 2 XX11 CHAR(1) INIT($04)
0860 DCL 1 OUT7,
0870 2 XX12 CHAR(4),
0880 2 XX13 CHAR(1) INIT($08),
0890 2 XX14 CHAR(1) INIT($08),
0900 2 XX15 CHAR(1) INIT($08),
0910 2 XX16 CHAR(1) INIT($08),
0920 2 XX17 CHAR(1) INIT($04)
0930 CARDCT=3
0940 TERMCT=3

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```

0950 CARDCT=$49
0960 TERMCT=$49
0970 COMMND:
0980 CALL DSPLY<, , ADDR(BUFEND)>
0990 $ LDX #PRMPT
1000 $ SCALL .DSPLZ
1010 CALL KEYIN<,8,ADDR(INPUT)> GIVING<,NUM>
1020 IF NUM EQ 0 THEN GO TO HELP
1030 PACK1=ADDR(INPUT)
1040 PACK1=PACK1+NUM
1050 DO WHILE NUM LE 7
1060 PACK1->ONE=" "
1070 PACK1=PACK1+1
1080 NUM=NUM+1
1090 END
1100 IF INPUT EQ CMD1 THEN GO TO ONLIN
1110 IF INPUT EQ CMD2 THEN GO TO ENDER
1120 IF INPUT EQ CMD3 THEN GO TO UPLOAD
1130 IF INPUT EQ CMD5 THEN GO TO HELP
1140 BUFFER="ERROR, ENTER "HELP" FOR HELP"
1150 CALL DSPLY<, , ADDR(BUFFER)>
1160 GO TO COMMND
1170 HELP: BUFFER="SUPPORTED COMMANDS ARE: "
1180 CALL DSPLY<, , ADDR(BUFFER)>
1190 BUFFER="ONLINE END UPLOAD"
1200 CALL DSPLY<, , ADDR(BUFFER)>
1210 GO TO COMMND
1220 ONLIN:
1230 BUFFER="GOING ONLINE THROUGH ACIA CARD"
1240 CALL DSPLY<, , ADDR(BUFFER)>
1250 BUFFER="ENTER "ESC" TO END ONLINE"
1260 CALL DSPLY<, , ADDR(BUFFER)>
1270 STATE=0
1280 TOP:
1290 CALL TERMIN
1300 IF NUM EQ 0 THEN GO TO CONT1
1310 IF ONE EQ ESCP THEN GO TO COMMND
1320 CALL CARDOU
1330 CONT1: CALL CARDIN
1340 IF NUM EQ 0 THEN GO TO TOP
1350 IF ONE EQ CNTN THEN STATE=1
1360 IF ONE EQ CNTO THEN STATE=0
1370 IF STATE EQ 1 THEN GO TO TOP
1380 CALL TERMOU
1390 GO TO TOP
1400 ENDER: CALL MDOS
1410 UPLOAD:
1420 BUFFER=OUT4
1430 CALL DSPLY<, , ADDR(BUFFER)>
1440 INPUT1=" "
1450 CALL KEYIN<,14,ADDR(INPUT1)> GIVING<,NUM>
1460 IF NUM EQ 0 THEN GO TO NOGO
1470 DUMMY=" "
1480 $ LDX #DUMMY
1490 $ STX PACK1
1500 PACK2=ADDR(UNIT)
1510 $ LDX #PACK1
1520 $ SCALL .PFNAM
1530 $ STAB STATUS
1540 IF STATUS NE 0
1550 THEN DO
1560 STATUS=7
1570 GO TO ERRUP
1580 END
1590 MODE=1
1600 TYPE="DK"
1610 $ LDX #IOCB
1620 $ SCALL .RESRV
1630 IF STATUS NE 0 THEN GO TO ERRUP
1640 BUFFS = ADDR(BUFF)
1650 BUFFE = 132+BUFFS
1660 FORMAT = 5
1670 ALLOC = 0
1680 SECTS = ADDR(SECT0)
1690 SECTE = 2559+SECTS
1700 $ LDX #IOCB
1710 $ SCALL .OPEN
1720 IF STATUS NE 0 THEN GO TO ERRUP
1730 CALL TERMIN
1740 CALL CARDIN
1750 $ LDX #OUT6
1760 $ SCALL .DSPLZ
1770 LNUM=1
1780 GO9:
1790 XX12=LNUM
1800 LNUM=LNUM+1
1801 GOX:
1810 $ LDX #IOCB
1820 $ SCALL .GETRC
1830 IF STATUS NE 0 THEN GO TO CLOSE
1840 $ LDX #OUT7
1850 $ SCALL .DSPLZ
1860 GO10: DO INDEX=1 TO 129
1870 ONE=BUFF(INDEX)
1871 IF ONE EQ BUFEND AND INDEX EQ 1 THEN GO TO GOX
1880 CALL CARDOU
1890 IF ONE EQ BUFEND THEN GO TO ENDRC
1900 END
1910 ENDRC: CALL INSYS
1920 IF NUM EQ 101 THEN GO TO CLOSE

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1930 GO TO GO9
1940 CLOSE:
1950 CALL DSPLY<, , ADDR(BUFEND)>
1960 $ LDX #IOCB
1970 $ SCALL .CLOSE
1980 IF STATUS NE 0 THEN GO TO ERRUP
1990 $ LDX #IOCB
2000 $ SCALL .RELES
2010 IF STATUS NE 0 THEN GO TO ERRUP
2020 IF NUM EQ 101 THEN GO TO NOGO
2030 BUFFER='UPLOAD COMPLETED'
2040 CALL DSPLY<, , ADDR(BUFFER)>
2050 STATE=0
2060 CALL CARDIN
2070 CALL TERMIN
2080 ONE=BUFEND
2090 CALL CARDOU
2100 GO TO TOP
2110 ERRUP:
2120 $ CLRB
2130 $ LDX #IOCB
2140 $ SCALL .MDERR
2150 NOGO: BUFFER='UPLOAD ABORTED'
2160 CALL DSPLY<, , ADDR(BUFFER)>
2170 GO TO TOP
2180 END
2190 TERMIN: PROC
2200 $ LDA A TERMCT
2210 $ ASR A
2220 $ BCC ENDTRM
2230 $ LDA A TERMTR
2240 $ STA A ONE
2250 NUM=1
2260 RETURN
2270 ENDTRM: NUM=0
2280 RETURN
2290 END
2300 CARDOU: PROC
2310 $ LDA A ONE
2320 $TRANS1 LDA B CARDCT
2330 $ BIT B #02H
2340 $ BEQ TRANS1
2350 $ STA A CARDTR
2360 RETURN
2370 END
2380 CARDIN: PROC
2390 $ LDA A CARDCT
2400 $ ASR A
2410 $ BCC ENDCRD
2420 $ LDA A CARDTR
2430 $ STA A ONE
2440 NUM=1
2450 RETURN
2460 ENDCRD: NUM=0
2470 RETURN
2480 END
2490 TERMOU: PROC
2500 $ LDA A ONE
2510 $TRANS2 LDA B TERMCT
2520 $ BIT B #02H
2530 $ BEQ TRANS2
2540 $ STA A TERMTR
2550 RETURN
2560 END
2570 INSYS: PROC
2580 LATCH3=60282
2590 LATCH2=65535
2600 CNTL3=#83
2610 CNTL2=#01
2620 CNTL1=#00
2630 PACK1=COUNT2
2640 TIME=0
2650 INDEX1=1
2660 BUFFER=' '
2670 DO WHILE TIME LE 10
2680 CALL CARDIN
2690 IF NUM NE 0 THEN DO
2700 IF ONE EQ CNTQ THEN GO TO GO1
2710 INDEX1=INDEX1+1
2720 IF INDEX1 EQ 60 THEN GO TO GO2
2730 END
2740 TIME=65535-COUNT2
2750 END
2760 GO2: BUFFER='COMMUNICATIONS TIMEOUT'
2770 CALL DSPLY<, , ADDR(BUFFER)>
2780 NUM=101
2790 RETURN
2800 GO1: NUM=0
2810 RETURN
2820 END

```

```

*****
*
* PROGRAM TO MAP THE RECORDED SPECTRAL CHARACTERISTICS TO THE
* FRAME OF REFERENCE OF THE POLYGRAPHIC PAPER RECORDING FOR
* CORRELATION WITH THE MANUALLY CLASSIFIED RECORD
*
*****
0010  CONVRT:
0020  PROC OPTIONS(MAIN)
0030  * NAM CONVRT
0040  DCL
0050      1 IOCB(2),
0060      2 STATUS BIN(1),
0070      2 MODE BIN(1),
0080      2 BUFPF BIN(2),
0090      2 BUFPF BIN(2),
0100      2 BUFE BIN(2),
0110      2 TYPE CHAR(2),
0120      2 FILE,
0130      3 UNIT CHAR(1),
0140      3 NAME CHAR(8),
0150      3 SUF CHAR(2),
0160      2 DUM0 BIN(2),
0170      2 FORMAT BIN(1),
0180      2 DUM1 BIN(3),
0190      2 DUM2 BIN(2),
0200      2 ALLOC BIN(2),
0210      2 SECTS BIN(2),
0220      2 SECTE BIN(2),
0230      2 SECTP BIN(2)
0240  DCL 1 TFILE,
0250      2 TUNIT CHAR(1),
0260      2 TNAME CHAR(8),
0270      2 TSUF CHAR(2)
0280  DCL DUMMY CHAR(1)
0290  DCL INPUT1 CHAR(13)
0300  DCL BUFFER CHAR(40)
0310  DCL BUFEEND CHAR(1) INIT(*D)
0320  DCL SECT0 CHAR(256)
0330  DCL SECT1 CHAR(256)
0340  DCL SECT2 CHAR(256)
0350  DCL SECT3 CHAR(256)
0360  DCL SECT4 CHAR(256)
0370  DCL SECT5 CHAR(256)
0380  DCL SECT6 CHAR(256)
0390  DCL SECT7 CHAR(256)
0400  DCL SECT8 CHAR(256)
0410  DCL SECT9 CHAR(256)
0420  DCL SECT10 CHAR(256)
0430  DCL SECT11 CHAR(256)
0440  DCL SECT12 CHAR(256)
0450  DCL SECT13 CHAR(256)
0460  DCL SECT14 CHAR(256)
0470  DCL SECT15 CHAR(256)
0480  DCL SECT16 CHAR(256)
0490  DCL SECT17 CHAR(256)
0500  DCL SECT18 CHAR(256)
0510  DCL SECT19 CHAR(256)
0520  DCL PAGE BIN(2)
0530  DCL SECS BIN(1)
0540  DCL NOWVAL BIN(2)
0550  DCL NUM1 BIN(1)
0560  DCL NUM2 BIN(1)
0570  DCL NUM10 BIN(1)
0580  DCL INDEX1 BIN(1)
0590  DCL PACK1 BIN(2)
0600  DCL PACK2 BIN(2)
0610  DCL ONE SIGNED BIN(1)
0620  DCL POINT BIN(2)
0630  DCL POINT1 BIN(2)
0640  DCL BUFP1(256)
0650  DCL NUM BIN(1)
0660  DCL INDEX2 BIN(1)
0670  DCL BUFOUT CHAR(45) INIT(' ')
0680  DCL HOLD CHAR(4)
0690  DCL PAGMAX BIN(2)
0700  DCL PAGMIN BIN(2)
0710  DCL PAGREF BIN(2)
0720  DCL NOGOC BIN(1)
0730  BUFFER='ENTER INPUT HEX FILE NAME'
0740  CALL DPLY<, , ADDR(BUFFER)>
0750  INPUT1=' '
0760  CALL KEYIN<, 13, ADDR(INPUT1)> GIVING<, NUM>
0770  IF NUM EQ 0 THEN GO TO NOGO
0780  DUMMY=' '
0790  * LDX #DUMMY
0800  * STX PACK1
0810  PACK2=ADDR(TFILE)
0820  * LDX #PACK1
0830  * SCALL .PFNAM
0840  * STAB NUM
0850  IF NUM NE 0
0860  THEN DO
0870      STATUS(1)=7
0880      GO TO ERRUP
0890  END
0900  FILE(1)=TFILE
0910  TYPE(1)='DK'
0920  TYPE(2)='DK'
0930  MODE(1)=1

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0940 MODE(2)=2
0950 $ LDX #IOCB
0960 $ SCALL .RESRV
0970 IF STATUS(1) NE 0 THEN GO TO ERRUP
0980 BUFFS(1)=ADDR(BUFF1)
0990 BUFFS(2)=ADDR(BUFOUT)
1000 BUFFE(1)=255+BUFFS(1)
1010 BUFFE(2)=44+BUFFS(2)
1020 FORMAT(1)=3
1030 FORMAT(2)=5
1040 ALLOC(1)=0
1050 ALLOC(2)=200
1060 SECTS(1)=ADDR(SECT0)
1070 SECTS(2)=ADDR(SECT10)
1080 SECTE(1)=2559+SECTS(1)
1090 SECTE(2)=2559+SECTS(2)
1100 $ LDX #IOCB
1110 $ SCALL .OPEN
1120 IF STATUS(1) NE 0 THEN GO TO ERRUP
1130 BUFFER='ENTER OUTPUT FILE NAME'
1140 CALL DSPLY< , ADDR(BUFFER)>
1150 INPUT1=' '
1160 CALL KEYIN< ,13, ADDR(INPUT1)> GIVING< ,NUM>
1170 IF NUM EQ 0 THEN GO TO NOGO
1180 DUMMY=' '
1190 $ LDX #DUMMY
1200 $ STX PACK1
1210 PACK2=ADDR(TFILE)
1220 $ LDX #PACK1
1230 $ SCALL .PFNAM
1240 $ STA B NUM
1250 IF NUM NE 0 THEN DO
1260 STATUS(1)=7
1270 GO TO ERRUP
1280 END
1290 FILE(2)=TFILE
1300 $ LDX #IOCB+37
1310 $ SCALL .RESRV
1320 IF STATUS(2) NE 0 THEN DO
1330 STATUS(1)=STATUS(2)
1340 GO TO ERRUP
1350 END
1360 $ LDX #IOCB+37
1370 $ SCALL .OPEN
1380 IF STATUS(2) NE 0 THEN DO
1390 STATUS(1)=STATUS(2)
1400 GO TO ERRUP
1410 END
1420 PACK1=ADDR(BUFOUT)
1430 INDEX2=1
1440 NOGOC=0
1450 NOWVAL=0
1460 NUM10=0
1470 NUM1=0
1480 BUFFER='ENTER STARTING PAGE #'
1490 CALL DSPLY< , ADDR(BUFFER)>
1500 CALL KEYIN< ,3, ADDR(INPUT1)> GIVING< ,NUM>
1510 IF NUM EQ 0 THEN GO TO NOGO
1520 PACK1=ADDR(INPUT1)
1530 DO WHILE NUM GE 1
1540 CALL HEXIN
1550 IF NOGOC EQ 1 THEN GO TO NOGO
1560 END
1570 PAGE=NOWVAL
1580 NOWVAL=0
1590 BUFFER='ENTER STARTING SEC ON PAGE'
1600 CALL DSPLY< , ADDR(BUFFER)>
1610 CALL KEYIN< ,2, ADDR(INPUT1)> GIVING< ,NUM>
1620 IF NUM EQ 0 THEN GO TO NOGO
1630 PACK1=ADDR(INPUT1)
1640 DO WHILE NUM GE 1
1650 CALL HEXIN
1660 IF NOGOC EQ 1 THEN GO TO NOGO
1670 END
1680 SECS=NOWVAL
1690 PAGMIN=0
1700 PAGMAX=30-SECS
1710 PAGREF=PAGMAX
1720 LOOP:
1730 POINT1=ADDR(BUFF1)
1740 $ LDX #IOCB
1750 $ SCALL .GETRC
1760 IF STATUS(1) EQ 9 THEN GO TO CLOSE
1770 IF STATUS(1) NE 0 THEN GO TO ERRUP
1780 CALL ASCII
1790 IF INDEX1 NE 0 THEN GO TO ERRUP
1800 GO TO LOOP
1810 CLOSE:
1820 IF INDEX2 NE 1 THEN DO
1830 $ LDX #IOCB+37
1840 $ SCALL .PUTRC
1850 IF STATUS(2) NE 0 THEN DO
1860 STATUS(1)=STATUS(2)
1870 GO TO ERRUP
1880 END
1890 END
1900 $ LDX #IOCB
1910 $ SCALL .CLOSE
1920 IF STATUS(1) NE 0 THEN GO TO ERRUP
1930 $ LDX #IOCB

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1940 $ SCALL .RELES
1950 IF STATUS(1) NE 0 THEN GO TO ERRUP
1960 $ LDX #IOCB+37
1970 $ SCALL .CLOSE
1980 IF STATUS(2) NE 0 THEN DO
1990     STATUS(1)=STATUS(2)
2000     GO TO ERRUP
2010 END
2020 $ LDX #IOCB+37
2030 $ SCALL .RELES
2040 IF STATUS(2) NE 0 THEN DO
2050     STATUS(1)=STATUS(2)
2060     GO TO ERRUP
2070 END
2080 BUFFER='CONVERSION COMPLETE'
2090 CALL DSPLY<, , ADDR(BUFFER)>
2100 CALL MDOS
2110 ERRUP:
2120 $ CLRB
2130 $ LDX #IOCB
2140 $ SCALL .MDERR
2150 NOGO: BUFFER='CONVERT ABORTED'
2160 CALL DSPLY<, , ADDR(BUFFER)>
2170 CALL MDOS
2180 END
2190 ASCII: PROC
2200     DO POINT=POINT1 TO BUFFP(1)
2210     IF INDEX2 EQ 1 THEN DO
2220         NUM1=POINT->NUM1
2230         INDEX2=INDEX2+1
2240         GO TO CONT1
2250     END
2260     IF INDEX2 EQ 2 THEN DO
2270         NUM2=POINT->NUM2
2280         IF NUM2 GT 150 AND NUM1 NE NUM10 THEN NUM1=NUM10
2290         NUM10=NUM1
2300         PACK2=ADDR(NUM1)
2310         PACK2=PACK2->PACK2
2320         DO WHILE PACK2 GT PAGMAX
2330             PAGE=PAGE+1
2340             PAGMAX=PAGMAX+30
2350             PAGMIN=PAGMAX-30
2360         END
2370         PACK2=PACK2-PAGMIN
2380         IF PAGMAX EQ PAGREF THEN PACK2=PACK2+SECS
2390         PACK1->HOLD=PAGE
2400         PACK1=PACK1+5
2410         PACK1->HOLD=PACK2
2420         PACK1=PACK1+5
2430         INDEX2=INDEX2+1
2440         GO TO CONT1
2450     END
2460     PACK1->HOLD=POINT->ONE
2470     PACK1=PACK1+5
2480     INDEX2=INDEX2+1
2490     IF INDEX2 EQ 10 THEN DO
2500         PACK1=ADDR(BUFOUT)
2510         INDEX2=1
2520 $     LDX #IOCB+37
2530 $     SCALL .PUTRC
2540     BUFOUT=' '
2550     IF STATUS(2) NE 0 THEN DO
2560         STATUS(1)=STATUS(2)
2570         INDEX1=99
2580         RETURN
2590     END
2600 END
2610 CONT1:
2620 END
2630 INDEX1=0
2640 RETURN
2650 END
2660 HEXIN: PROC
2670     PACK2=NOWVAL
2680     DO INDEX1=1 TO 9
2690         NOWVAL=NOWVAL+PACK2
2700     END
2710     DUMMY=PACK1->DUMMY
2720 $     LDA A DUMMY
2730 $     SCALL .NUMD
2740 $     BCS NO
2750 $     STA A NUM1
2760     NOWVAL=NOWVAL + NUM1
2770     PACK1=PACK1+1
2780     NUM=NUM-1
2790     RETURN
2800 NO: NOGOC=1
2810     RETURN
2820 END

```

```

*****
*
* PROGRAM TO CONTROL THE RECORDING OF THE SPECTRAL ESTIMATES
* GENERATED BY THE 2920S
*
*****
0010 RECORD: PROC OPTIONS(MAIN)
0020 $ NAM RECORD
0030 DCL 1 IOCB,
0040 2 STATUS BIN(1),
0050 2 MODE BIN(1),
0060 2 BUFFP BIN(2),
0070 2 BUFFS BIN(2),
0080 2 BUFEE BIN(2),
0090 2 TYPE CHAR(2) INIT('DK'),
0100 2 UNIT CHAR(1),
0110 2 NAME CHAR(8),
0120 2 SUF CHAR(2),
0130 2 DUM0 BIN(2),
0140 2 FORMAT BIN(1),
0150 2 DUM1 BIN(3),
0160 2 DUM2 BIN(2),
0170 2 ALLOC BIN(2),
0180 2 SECTS BIN(2),
0190 2 SECTE BIN(2),
0200 2 SECTP BIN(2)
0210 DCL 1 BUFF0(100),
0220 2 BUFF1(256) BIN(1)
0230 DCL CNTL BIN(1)
0240 DCL IRQST LABEL INIT(INDATA)
0250 DCL SECTOR BIN(128)
0260 DCL BUFFER CHAR(40)
0270 DCL BUFEND CHAR(1) INIT($D)
0280 DCL 1 OUT,
0290 2 PRT1 CHAR(22) INIT('TRANSITIONS DETECTED: '),
0300 2 PRT2 CHAR(5),
0310 2 PRT3 CHAR(1) INIT($04)
0320 DCL 1 BACK,
0330 2 PRT4 CHAR(1) INIT($08),
0340 2 PRT5 CHAR(1) INIT($08),
0350 2 PRT6 CHAR(1) INIT($08),
0360 2 PRT7 CHAR(1) INIT($08),
0370 2 PRT8 CHAR(1) INIT($08),
0380 2 PRT9 CHAR(5),
0390 2 PRT0 CHAR(1) INIT($04)
0400 DCL INPUT CHAR(1)
0410 DCL PACK1 BIN(2)
0420 DCL PACK2 BIN(2)
0430 DCL INDEX1 BIN(1)
0440 DCL NUM BIN(1)
0450 DCL NUM1 SIGNED BIN(1)
0460 DCL ONE CHAR(1)
0470 DCL PAST(7) BIN(1)
0480 DCL NOW(7) BIN(1)
0490 DCL SENSE(7) SIGNED BIN(1)
0500 DCL POINT3 BIN(2)
0510 DCL TRANS BIN(2)
0520 DCL DIFF SIGNED BIN(1)
0530 DCL XX SIGNED BIN(1)
0540 DCL YY SIGNED BIN(1)
0550 DCL ZZ SIGNED BIN(1)
0560 DCL POINT BIN(2)
0570 DCL POINT0 BIN(2)
0580 DCL POINT1 BIN(2)
0590 DCL POINT2 BIN(2)
0600 DCL COUNT2 BIN(2) DEF $E03C
0610 DCL LATCH3 BIN(2) DEF $E03E
0620 DCL LATCH2 BIN(2) DEF $E03C
0630 DCL CNTL3 BIN(1) DEF $E038
0640 DCL CNTL2 BIN(1) DEF $E039
0650 DCL CNTL1 BIN(1) DEF $E038
0660 DCL TERMCT BIN(1) DEF $FCF4
0670 DCL TERMTR BIN(1) DEF $FCF5
0680 DCL PIA1AD BIN(1) DEF $E010
0690 DCL PIA1AC BIN(1) DEF $E011
0700 DCL PIA1BD BIN(1) DEF $E012
0710 DCL PIA1BC BIN(1) DEF $E013
0720 DCL PIA2AD BIN(1) DEF $E020
0730 DCL PIA2AC BIN(1) DEF $E021
0740 DCL PIA2BD BIN(1) DEF $E022
0750 DCL PIA2BC BIN(1) DEF $E023
0760 DCL PIA3AD BIN(1) DEF $E030
0770 DCL PIA3AC BIN(1) DEF $E031
0780 DCL PIA3BD BIN(1) DEF $E032
0790 DCL PIA3BC BIN(1) DEF $E033
0800 DCL PIA4AD BIN(1) DEF $E034
0810 DCL PIA4AC BIN(1) DEF $E035
0820 DCL PIA4BD BIN(1) DEF $E036
0830 DCL PIA4BC BIN(1) DEF $E037
0840 $ LDX CBUF$
0850 $ STX PACK1
0860 PACK2=ADDR(UNIT)
0870 $ LDX $PACK1
0880 $ SCALL , PFNAM
0890 $ STA B STATUS
0900 IF STATUS GT 1 THEN DO
0910 STATUS=7
0920 GO TO ERR
0930 END
0940 PIA1AD=0

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0950 PIA1BD=0
0960 PIA2AD=0
0970 PIA2BD=0
0980 PIA3AD=0
0990 PIA3BD=0
1000 PIA4AD=0
1010 PIA4BD=0
1020 PIA1AC=4
1030 PIA1BC=4
1040 PIA2AC=4
1050 2BC=4
1060 PIA3AC=4
1070 PIA3BC=4
1080 PIA4AC=4
1090 PIA4BC=4
1100 TERMCT=$03
1110 TERMCT=$49
1120 $ LDX IRQST
1130 $ STX IRQ$UV
1140 MODE=2
1150 BUFFS=ADDR(BUFF0)
1160 BUFFE=99+BUFFS
1170 FORMAT=3
1180 ALLOC=100
1190 SECTS=ADDR(SECTOR)
1200 SECTE=127+SECTS
1210 $ LDX $IOCB
1220 $ SCALL .RESRV
1230 IF STATUS NE 0 THEN GO TO ERR
1240 $ LDX $IOCB
1250 $ SCALL .OPEN
1260 IF STATUS NE 0 THEN GO TO ERR
1270 POINT1=ADDR(BUFF0)
1280 POINT2=POINT1+25599
1290 CNTL=0
1300 TRANS=0
1310 POINT0=ADDR(PAST)
1320 DO INDEX1=1 TO 7
1330 POINT0->NUM=0
1340 POINT0=POINT0+1
1350 END
1360 SENSE(1)=3
1370 SENSE(2)=3
1380 SENSE(3)=3
1390 SENSE(4)=3
1400 SENSE(5)=3
1410 SENSE(6)=3
1420 SENSE(7)=25
1430 BUFFER='ONCE RECORDING HAS BEGUN,'
1440 CALL DSPLY< , , ADDR(BUFFER)>
1450 BUFFER='ENTER ESC TO TERMINATE'
1460 CALL DSPLY< , , ADDR(BUFFER)>
1470 BUFFER='RETURN TO START RECORDING'
1480 CALL DSPLY< , , ADDR(BUFFER)>
1490 $ LDA B $00H
1500 $ LDX $INPUT
1510 $ SCALL .KEYIN
1520 LATCH3=60282
1530 LATCH2=65535
1540 CNTL3=$83
1550 CNTL2=$01
1560 CNTL1=$00
1570 PACK1=COUNT2
1580 PRT2=0
1590 $ LDX $OUT
1600 $ SCALL .DSPLZ
1610 PIA1AC=12
1620 LOOP:
1630 $ LDA A TERMCT
1640 $ ASR A
1650 $ BCC NOIN
1660 PIA1AC=4
1670 BUFFER=' '
1680 CALL DSPLY< , , ADDR(BUFFER)>
1690 GO TO OUTREQ
1700 NOIN: IF CNTL NE 0 THEN GO TO OUTNOW
1710 GO TO LOOP
1720 OUTNOW: PIA1AC=4
1730 BUFFER=' '
1740 CALL DSPLY< , , ADDR(BUFFER)>
1750 BUFFER='MEMORY LIMIT REACHED'
1760 CALL DSPLY< , , ADDR(BUFFER)>
1770 OUTREQ: BUFFER='RECORDING TERMINATED'
1780 CALL DSPLY< , , ADDR(BUFFER)>
1790 POINT0=BUFFS
1800 DO WHILE BUFFE LE POINT1
1810 $ LDX $IOCB
1820 $ SCALL .PUTRC
1830 IF STATUS NE 0 THEN GO TO ERR
1840 BUFFS=BUFFS+100
1850 BUFFE=BUFFS+99
1860 END
1870 IF BUFFE NE POINT1 THEN DO
1880 BUFFE=POINT1
1890 $ LDX $IOCB
1900 $ SCALL .PUTRC
1910 IF STATUS NE 0 THEN GO TO ERR
1920 END
1930 $ LDX $IOCB
1940 $ SCALL .CLOSE

```

```

1950 IF STATUS NE 0 THEN GO TO ERR
1960 $ LDX #IOCB
1970 $ SCALL .RELES
1980 IF STATUS NE 0 THEN GO TO ERR
1990 CALL MDOS
2000 ERR:
2010 $ CLRB
2020 $ LDX #IOCB
2030 $ SCALL .MDERR
2040 CALL MDOS
2050 END
2060 INDATA: PROC
2070 POINT=ADDR(NOW)
2080 POINT0=ADDR(PAST)
2090 POINT3=ADDR(SENSE)
2100 NOW(1)=PIA1AD
2110 NOW(2)=PIA1BD
2120 NOW(3)=PIA2AD
2130 NOW(4)=PIA2BD
2140 NOW(5)=PIA3AD
2150 NOW(6)=PIA3BD
2160 NOW(7)=PIA4AD
2170 DO INDEX1=1 TO 7
2180 XX=POINT->NUM1
2190 YY=POINT0->NUM1
2200 ZZ=POINT3->NUM1
2210 DIFF=0
2220 IF XX GT YY THEN DIFF=XX-YY
2230 IF YY GT XX THEN DIFF=YY-XX
2240 IF DIFF GE ZZ THEN GO TO CHANGE
2250 POINT=POINT+1
2260 POINT0=POINT0+1
2270 POINT3=POINT3+1
2280 END
2290 $ RTI
2300 CHANGE: POINT=ADDR(NOW)
2310 POINT0=ADDR(PAST)
2320 POINT1->PACK1=G5535-COUNT2
2330 POINT1=POINT1+2
2340 DO INDEX1=1 TO 7
2350 POINT0->NUM=POINT->NUM
2360 POINT1->NUM=POINT->NUM
2370 POINT=POINT+1
2380 POINT0=POINT0+1
2390 POINT1=POINT1+1
2400 END
2410 TRANS=TRANS+1
2420 PRT9=TRANS
2430 $ LDX #BACK
2440 $ SCALL .DSPLZ
2450 PACK1=POINT2-POINT1
2460 IF PACK1 LT 9 THEN CNTL=1
2470 $ RTI
2480 END

```

```

*****
*
* PROGRAM TO VERIFY CORRECT OPERATION OF 2920S ON POWERUP
*
*****
0010 POWER: PROC OPTIONS(MAIN)
0020 $ NAM POWER
0030 DCL IRGST LABEL INIT(INDATA)
0040 DCL OUT CHAR(45)
0050 DCL OUTEND CHAR(1) INIT(0)
0060 DCL INDEX1 BIN(1)
0070 DCL NUM1 SIGNED BIN(3)
0080 DCL ONE1 CHAR(5)
0090 DCL NOW(7) BIN(1)
0100 DCL POINT BIN(2)
0110 DCL POINT1 BIN(2)
0120 DCL TERMCT BIN(1) DEF $FCF4
0130 DCL TERMTR BIN(1) DEF $FCF5
0140 DCL PIA1AD BIN(1) DEF $E010
0150 DCL PIA1AC BIN(1) DEF $E011
0160 DCL PIA1BD BIN(1) DEF $E012
0170 DCL PIA1BC BIN(1) DEF $E013
0180 DCL PIA2AD BIN(1) DEF $E020
0190 DCL PIA2AC BIN(1) DEF $E021
0200 DCL PIA2BD BIN(1) DEF $E022
0210 DCL PIA2BC BIN(1) DEF $E023
0220 DCL PIA3AD BIN(1) DEF $E030
0230 DCL PIA3AC BIN(1) DEF $E031
0240 DCL PIA3BD BIN(1) DEF $E032
0250 DCL PIA3BC BIN(1) DEF $E033
0260 DCL PIA4AD BIN(1) DEF $E034
0270 DCL PIA4AC BIN(1) DEF $E035
0280 DCL PIA4BD BIN(1) DEF $E036
0290 DCL PIA4BC BIN(1) DEF $E037
0300 PIA1AD=0
0310 PIA1BD=0
0320 PIA2AD=0
0330 PIA2BD=0
0340 PIA3AD=0
0350 PIA3BD=0
0360 PIA4AD=0
0370 PIA4BD=0
0380 PIA1AC=4
0390 PIA1BC=4
0400 PIA2AC=4
0410 PIA2BC=4
0420 PIA3AC=4
0430 PIA3BC=4
0440 PIA4AC=4
0450 PIA4BC=4
0460 TERMCT=403
0470 TERMTR=449
0480 OUT= ' '
0490 $ LDX IRGST
0500 $ STX IRG1UV
0510 PIA1AC=12
0520 LOOP:
0530 $ LDA A TERMCT
0540 $ ASR A
0550 $ BCC LOOP
0560 PIA1AC=4
0570 CALL MDOS
0580 END
0590 INDATA: PROC
0600 POINT=ADDR(NOW)
0610 NOW(1)=PIA1AD
0620 NOW(2)=PIA1BD
0630 NOW(3)=PIA2AD
0640 NOW(4)=PIA2BD
0650 NOW(5)=PIA3AD
0660 NOW(6)=PIA3BD
0670 NOW(7)=PIA4AD
0680 POINT1=ADDR(OUT)
0690 DO INDEX1=1 TO 7
0700 POINT1->ONE1=POINT->NUM1
0710 POINT1=POINT1+5
0720 POINT=POINT+1
0730 END
0740 CALL DSPLY<, , ADDR(OUT)>
0750 $ RTI
0760 END

```



```

*****
*
* PROGRAM TO ASSEMBLE 2920 PROGRAMS
*
*****
0010 AS2920: PROC OPTIONS(MAIN) /* MAINLINE ROUTINE */
0020 $ NAM AS2920 /*
0030 DECLARE /*
0040 1 IOCB(3), /* FILE I/O CONTROL BLOCKS */
0050 2 STATUS BIN (1), /* RETURN CODE */
0060 2 MODE BIN (1), /* INPUT, OUTPUT, OR UPDATE */
0070 2 BUFPF BIN (2), /* RECORD BUFFER POINTER */
0080 2 BUFPF BIN (2), /* RECORD BUFFER START ADDR */
0090 2 BUFE BIN (2), /* RECORD BUFFER END ADDRESS */
0100 2 TYPE CHAR(2), /* DEVICE TYPE */
0110 2 FILE, /* FILE NAME STRUCTURE */
0120 3 UNIT CHAR(1), /* LOGICAL UNIT NUMBER */
0130 3 NAME CHAR(8), /* FILE NAME */
0140 3 SUF CHAR(2), /* FILE SUFFIX */
0150 2 DUM0 BIN (2), /* RESERVED */
0160 2 FORMAT BIN (1), /* BIN, ASCII, OR ASCII CONV */
0170 2 DUM1 BIN (5), /* RESERVED */
0180 2 ALLOC BIN (2), /* ALLOCATION INCREMENT */
0190 2 SECTS BIN (2), /* SECTOR BUFFER START ADDR */
0200 2 SECTE BIN (2), /* SECTOR BUFFER END ADDRESS */
0210 2 SECTP BIN (2) /* SECTOR BUFFER POINTER */
0220 DECLARE /*
0230 1 TFILE, /* TEMP FILE NAME STRUCTURE */
0240 2 TUNIT CHAR(1), /* LOGICAL UNIT NUMBER */
0250 2 TNAME CHAR(8), /* FILE NAME */
0260 2 TSUF CHAR(2) /* FILE SUFFIX */
0270 DECLARE /*
0280 1 CNTL, /* ASSEMBLY CONTROL */
0290 2 PRINT BIN (1), /* LISTING PRINT CONTROL */
0300 2 OBJECT BIN (1), /* OBJECT FILE CONTROL */
0310 2 PAGEL BIN (1), /* PAGE LENGTH CONTROL */
0320 2 INST BIN (1) INIT(0), /* EPROM INSTRUCTION COUNTER */
0330 2 LINE BIN (1), /* LISTING LINE COUNTER */
0340 2 PAGE BIN (1) INIT(1), /* PAGE COUNTER */
0350 2 ERROR BIN (1) INIT(0), /* ERROR COUNT */
0360 2 RTRN BIN (1), /* RETURN CODE */
0370 2 KIND BIN (1), /* TYPE OF TOKEN DECODED */
0380 2 RTRNT BIN (1), /* TOKEN PARSER RETURN CODE */
0390 2 VAR BIN (1) INIT(0), /* SYMBOL TABLE POINTER */
0400 2 STATE BIN (1), /* STATE INDICATOR */
0410 2 ENDYET BIN (1) /* SCAN REMAINDER OF RECORD? */
0420 DCL BUFFER CHAR(78) /* RECORD I/O BUFFER */
0430 DCL BUFEND CHAR(1) INIT($D) /* CR MARKER */
0440 DCL BUFEOT CHAR(1) INIT($H) /* EOT CHARACTER */
0450 DCL SECT1 CHAR(128) /* INPUT FILE SECTOR BUFFER */
0460 DCL SECT2 CHAR(256) /* LISTING FILE SECTOR BUFFER */
0470 DCL SECT3 CHAR(128) /* OBJECT FILE SECTOR BUFFER */
0480 DCL PARSE1 BIN (2) /* MDOS PARSER BUFFER */
0490 DCL PARSE2 BIN (2) /* MDOS PARSER BUFFER */
0500 DCL EPROMP BIN (2) /* EPROM POINTER */
0510 DECLARE /*
0520 1 EPROM(192), /* EPROM MEMORY SPACE */
0530 2 SEGE1 BIN(1), /* MOST SIGNIFICANT BYTE */
0540 2 SEGE2 BIN(1), /* MIDDLE BYTE */
0550 2 SEGE3 BIN(1) /* LEAST SIGNIFICANT BYTE */
0560 DECLARE /*
0570 1 BUFFO, /* OBJECT FILE I/O STRUCTURE */
0580 2 SEG1 BIN(1), /* MOST SIGNIFICANT BYTE */
0590 2 SEG2 BIN(1), /* MIDDLE BYTE */
0600 2 SEG3 BIN(1) /* LEAST SIGNIFICANT BYTE */
0610 DCL RAMNAM (40) CHAR(8) INIT(' ') /* RAM SYMBOL TABLE */
0620 DCL ENDRAM CHAR(1) INIT($D)
0630 DCL INDEX1 BIN (1) /* GENERAL UTILITY COUNTER */
0640 DCL INDEX2 BIN (1) /* GENERAL UTILITY COUNTER */
0650 DCL INDEX3 BIN (1) /* GENERAL UTILITY COUNTER */
0660 DCL NUM BIN (1) /* GENERAL UTILITY BIN(1) */
0670 DCL BUFF(80) CHAR(1) /* GENERAL CHAR BY 1 BUFFER */
0680 DCL BUFF1 CHAR(80) /* INPUT FILE BUFFER */
0690 DCL BUFFL CHAR(132) /* LISTING FILE I/O BUFFER */
0700 DCL LAST BIN (2) /* GENERAL UTILITY POINTER */
0710 DCL ONE CHAR(1) /* GENERAL UTILITY CHAR */
0720 DECLARE /*
0730 1 FOUR, /* FOUR CHAR STRUCTURE */
0740 2 FOURX (4) CHAR(1) /* ADDRESSABLE BY ONE */
0750 DECLARE /*
0760 1 HEADER, /* ASSEMBLER TITLES */
0770 2 CPT0 CHAR(1) INIT($C), /* FORM FEED CHARACTER */
0780 2 HEAD1, /* TOP OF PAGE MESSAGE */
0790 3 CPT1 CHAR(31) INIT('MDOS 2920 ASSEMBLER VER 1.00'),
0800 3 CPT20 CHAR(12) INIT(' '),
0810 3 CPT21 CHAR(8),
0820 3 CPT22 CHAR(1) INIT(' '),
0830 3 CPT23 CHAR(2),
0840 3 CPT24 CHAR(1) INIT(' '),
0850 3 CPT25 CHAR(1),
0860 3 CPT26 CHAR(16) INIT(' '),
0870 3 CPT3 CHAR(5) INIT('PAGE '),
0880 3 CPT4 CHAR(2),
0890 3 CPT5 CHAR(1) INIT($D)
0900 DECLARE /*
0910 1 HEAD2, /* SECOND LINE OF HEADER */
0920 2 CPT14 CHAR(27) INIT('LOC OBJECT SOURCE STATEMENT'),
0930 2 CPT15 CHAR(1) INIT($D),
0940 2 CPT16 CHAR(1) INIT($D)
0950 DECLARE /*

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0960      1      OUT,                                /* GENERAL OUTPUT LINE SHELL */
0970      2      LOC      CHAR(3),                  /* MEMORY LOCATION */
0980      2      CPT12     CHAR(1) INIT(' '),        /* */
0990      2      INSTL     CHAR(6) INIT(' '),        /* INSTRUCTION (HEX) SPACE */
1000      2      CPT13     CHAR(1) INIT(' '),        /* */
1010      2      ININST    CHAR(80)                 /* SUPPLIED SOURCE STATEMENT */
1020 DCL THREE BIN(3)                               /* UTILITY 3 BYTE VARIABLE */
1030 DECLARE                                         /* */
1040      1      MESSAGE,                             /* GENERAL MESSAGE STRUCTURE */
1050      2      P0      CHAR(40),                   /* MESSAGE PART */
1060      2      P1      CHAR(1) INIT('$D')          /* LINE FEED/CR */
1070 DECLARE                                         /* */
1080      1      TOTALS,                               /* END OF ASSEMBLY TOTAL ERRORS */
1090      2      MESS0     CHAR(14) INIT('TOTAL ERRORS: '),
1100      2      NUMPRT    CHAR(3),                  /* NUMBER OF ERRORS */
1110      2      ENDPRT    CHAR(1) INIT('$D')        /* CR */
1120 DCL MESS1 CHAR(40) INIT('* ERROR: TOKEN OVER 8 CHAR')
1130 DCL MESS2 CHAR(40) INIT('* ERROR: INVALID SYNTAX')
1140 DCL MESS3 CHAR(40) INIT('* ERROR: OUT AFTER DAR AS DEST')
1150 DCL MESS4 CHAR(40) INIT('* ERROR: OUT AFTER COND SUB')
1160 DCL MESS5 CHAR(40) INIT('* ERROR: CVT AFTER DAR AS DEST')
1170 DCL MESS6 CHAR(40) INIT('* ERROR: CVT AFTER CVT')
1180 DCL MESS7 CHAR(40) INIT('* ERROR: CVT AFTER COND SUB')
1190 DCL MESS8 CHAR(40) INIT('* ERROR: CVT WITH DAR AS DEST')
1200 DCL MESS9 CHAR(40) INIT('* ERROR: CND USED WITH AND')
1210 DCL MESS10 CHAR(40) INIT('* ERROR: CND USED WITH LIM')
1220 DCL MESS11 CHAR(40) INIT('* ERROR: CND USED WITH ABS')
1230 DCL MESS12 CHAR(40) INIT('* ERROR: CND/SUB & DAR AS DEST')
1240 DCL MESS13 CHAR(40) INIT('* ERROR: LIM WITH SHIFT NOT R0')
1250 DCL MESS14 CHAR(40) INIT('* ERROR: MORE THAN 40 VAR USED')
1260 DCL MESS15 CHAR(40) INIT('* ERROR: END-EQU NOT ALLOWED')
1270 DCL MESS16 CHAR(40) INIT('* ERROR: INVALID VAR NAME')
1280 DCL DAROUT(2) BIN(1)
1290 DCL CONSUB(2) BIN(1)
1300 DCL CVTX(2) BIN(1)
1310 DCL ANDX BIN(1)
1320 DCL LIMX BIN(1)
1330 DCL ABSX BIN(1)
1340 DCL SUBX BIN(1)
1350 DECLARE                                         /* */
1360      1      SCREEN,                               /* */
1370      2      CPT6     CHAR(30) INIT('MDOS 2920 ASSEMBLER VER 1.00'),
1380      2      CPT7     CHAR(1) INIT('$D'),
1390      2      CPT10    CHAR(1) INIT('$D'),
1400      2      CPT11    CHAR(1) INIT('$4')
1410 DECLARE                                         /* */
1420      1      TOKENA,                             /* TOKEN STRUCTURE */
1430      2      TOKEN (8) CHAR(1)                 /* ADDRESSABLE BY 1 */
1440 DECLARE                                         /* */
1450      1      TOP,                                   /* 24 BIT INSTRUCTION STRUCT */
1460      2      ADF0      BIT (1),                  /* I/O CODE */
1470      2      ADK2      BIT (1),                  /* I/O CODE */
1480      2      ADK1      BIT (1),                  /* I/O CODE */
1490      2      ADK0      BIT (1),                  /* I/O CODE */
1500      2      A2        BIT (1),                  /* SOURCE FIELD OPERAND */
1510      2      B1        BIT (1),                  /* DESTINATION FIELD OPERAND */
1520      2      A1        BIT (1),                  /* SOURCE FIELD OPERAND */
1530      2      ADF1      BIT (1),                  /* I/O CODE */
1540      2      A4        BIT (1),                  /* SOURCE FIELD OPERAND */
1550      2      B3        BIT (1),                  /* DESTINATION FIELD OPERAND */
1560      2      A3        BIT (1),                  /* SOURCE FIELD OPERAND */
1570      2      B2        BIT (1),                  /* DESTINATION FIELD OPERAND */
1580      2      A0        BIT (1),                  /* SOURCE FIELD OPERAND */
1590      2      B5        BIT (1),                  /* DESTINATION FIELD OPERAND */
1600      2      A5        BIT (1),                  /* SOURCE FIELD OPERAND */
1610      2      B4        BIT (1),                  /* DESTINATION FIELD OPERAND */
1620      2      S2        BIT (1),                  /* SHIFT CODE */
1630      2      S1        BIT (1),                  /* SHIFT CODE */
1640      2      S0        BIT (1),                  /* SHIFT CODE */
1650      2      B0        BIT (1),                  /* DESTINATION FIELD OPERAND */
1660      2      L2        BIT (1),                  /* OP CODE */
1670      2      L1        BIT (1),                  /* OP CODE */
1680      2      L0        BIT (1),                  /* OP CODE */
1690      2      S3        BIT (1),                  /* SHIFT CODE */
1700 DECLARE                                         /* */
1710      1      BITUP,                               /* BYTE ADDRESSABLE BY BIT */
1720      2      BIT7      BIT (1),                  /* MS BIT */
1730      2      BIT6      BIT (1),
1740      2      BIT5      BIT (1),
1750      2      BIT4      BIT (1),
1760      2      BIT3      BIT (1),
1770      2      BIT2      BIT (1),
1780      2      BIT1      BIT (1),
1790      2      BIT0      BIT (1),
1800 DECLARE                                         /* LS BIT */
1810      1      BREAK,                               /* BY ONE CHAR ADDRESSABLE */
1820      2      CHAR6     CHAR (1),                  /* MOST SIGNIFICANT CHAR */
1830      2      CHAR5     CHAR (1),
1840      2      CHAR4     CHAR (1),
1850      2      CHAR3     CHAR (1),
1860      2      CHAR2     CHAR (1),
1870      2      CHAR1     CHAR (1),
1880 DCL OPCODE (87) CHAR(4)                        /* TABLE OF OPCODES */
1890 INIT('XOR','AND','LIM','ABS','ABA','SUB','ADD','LDA',
1900      'IN0','IN1','IN2','IN3','NOP','EOP','CVT5','CND5',
1910      'OUT0','OUT1','OUT2','OUT3','OUT4','OUT5','OUT6','OUT7',
1920      'CVT0','CVT1','CVT2','CVT3','CVT4','CVT5','CVT6','CVT7',
1930      'CND0','CND1','CND2','CND3','CND4','CND5','CND6','CND7',
1940      'R01','R02','R03','R04','R05','R06','R07','R08','R09','R10',
1950      'R11','R12','R13','L02','L01','R00','R0','R1','R2','R3','R4',

```



```

2960      CALL OPTION GIVING<OBJECT, ,>
2970      GO TO (LOOP1,ERR,ERR7),RTRN
2980  END
2990  IF PARSE1-> ONE EQ 'L' THEN DO
3000      INDEX3=2
3010      OFFSET=37
3020      TSUF='AL'
3030      CALL OPTION GIVING<PRINT, ,>
3040      GO TO (LOOP1,ERR,ERR7), RTRN
3050  END
3060  GO TO ERR
3070  END
3080  GO TO CONT2
3090  ERR: BUFFER='*** INVALID OPTIONS ***'
3100      CALL D$PLY<, , ADDR(BUFFER)>
3110      INDEX1=0
3120  ERR7: IF OBJECT NE 1 THEN GO TO ERR5
3130  $ LDX #IOCB+74
3140  $ SCALL .CLOSE
3150  $ LDX #IOCB+74
3160  $ SCALL .RELES
3170  ERR5: IF PRINT NE 1 THEN GO TO ERR3
3180  $ LDX #IOCB+37
3190  $ SCALL .CLOSE
3200  $ LDX #IOCB+37
3210  $ SCALL .RELES
3220  ERR3:
3230  $ LDX #IOCB
3240  $ SCALL .CLOSE
3250  ERR2:
3260  $ LDX #IOCB
3270  $ SCALL .RELES
3280  ERR1: STATUS(1)=INDEX1
3290      IF STATUS(1) EQ 0 THEN CALL MDOS
3300  $ CLR
3310  $ LDX #IOCB
3320  $ SCALL .MDERR
3330      CALL MDOS
3340  CONT2:
3350  $ LDX #SCREEN
3360  $ SCALL .D$PLZ
3370      SEG1=$40
3380      SEG2=$00
3390      SEG3=$EF
3400      DO INDEX1=1 TO 192
3410          EPROM(INDEX1)=BUFFO
3420      END
3430      CALL TOPPG
3440      IF RTRN NE 0 THEN GO TO ERR7
3450  LOOP2:
3460      SEG1=$40
3470      SEG2=$00
3480      SEG3=$EF
3490      ENDYET=0
3500      STATE=1
3510      RTRNT=1
3520      PARSE1=ADDR(TOP)
3530      PARSE1->NUM=SEG1
3540      PARSE1=PARSE1+1
3550      PARSE1->NUM=SEG2
3560      PARSE1=PARSE1+1
3570      PARSE1->NUM=SEG3
3580      DAROUT(2)=DAROUT(1)
3590      DAROUT(1)=0
3600      CONSUB(2)=CONSUB(1)
3610      CONSUB(1)=0
3620      CVTX(2)=CVTX(1)
3630      CVTX(1)=0
3640      ANDX=0
3650      LIMX=0
3660  $ LDX #IOCB
3670  $ SCALL .GETRC
3680      ABSX=0
3690      SUBX=0
3700      IF STATUS(1) EQ 0 THEN GO TO CONT3
3710      IF STATUS(1) EQ 9 THEN GO TO EOF
3720      INDEX1=STATUS(1)
3730      BUFFER='ERROR IN READING INPUT FILE'
3740      CALL D$PLY<, , ADDR(BUFFER)>
3750      GO TO ERR7
3760  CONT3:
3770      PARSE1=ADDR(BUFFI)
3780      ININST=BUFFI
3790      IF PARSE1->ONE EQ '*' THEN GO TO COMMND
3800      NUM=PARSE1->NUM
3810      IF NUM GE 48 AND NUM LE 57 THEN DO
3820  LOOP9: PARSE1=PARSE1+1
3830      ONE=PARSE1->ONE
3840      IF ONE EQ ' ' THEN GO TO CONT5
3850      IF ONE EQ BUFEND THEN GO TO EOR
3860      GO TO LOOP9
3870  END
3880      GO TO LOOP3
3890  CONT5: PARSE1=PARSE1+1
3900  LOOP3: CALL PARSE
3910  GO TO (EOR,CALPRO,CALPRO,CALPRO,EORF,CALPRO,EORF),RTRNT
3920  CALPRO: CALL PROCES
3930      IF STATE EQ 8 THEN GO TO EORF
3940      IF ENDYET EQ 1 THEN GO TO EOR
3950      IF ENDYET EQ 2 THEN GO TO EORF

```

```

/* PROCESS OPTIONS */
/* CHECK RETURN CODE */
/* END OBJECT FILE PROCESSING */
/* LISTING FILE OPTION ? */
/* INDICATE PRINT FILE */
/* SET IOCB OFFSET */
/* SET DEFAULT FILE TYPE */
/* PROCESS OPTIONS */
/* CHECK RETURN CODE */
/* END PRINT FILE PROCESSING */
/* IF NOT O,L,CR,OR ' '-ERROR */
/* END WHILE */
/* IF DROPPED THROUGH WHILE-OK */
/* SET MESSAGE FOR OPTIONS NOT */
/* UNDERSTOOD AND DISPLAY */
/* SET MSG NUMBER FOR MDERR */
/* IF OBJECT FILE NOT OPENED */
/* POINT TO OBJECT IOCB */
/* AND CLOSE FILE */
/* POINT TO OBJECT IOCB */
/* RELEASE FILE */
/* IF NO PRINT FILE - BYPASS */
/* ELSE : POINT TO PRINT IOCB */
/* CLOSE PRINT FILE */
/* POINT TO PRINT IOCB */
/* RELEASE FILE */
/* ENTRY TO CLOSE INPUT FILE */
/* POINT TO INPUT FILE IOCB */
/* AND CLOSE FILE */
/* ENTRY POINT TO RELES INPUT */
/* POINT TO INPUT FILE IOCB */
/* AND RELEASE */
/* LOAD ERROR CODE */
/* IF NO MDERR ERROR - STOP */
/* CLEAR ACCB FOR MDERR */
/* POINT TO RETURN CODE */
/* DISPLAY ERROR MESSAGE */
/* AND STOP PROCESSING */
/* CONTINUATION POINT */
/* LOAD X WITH START OF MESSAGE */
/* AND DISPLAY ON CONSOLE */
/* SET DEFAULT EPROM VALUES */
/* */
/* */
/* */
/* PRINT TOP PAGE HEADING */
/* IF I/O PROBLEM-MOPUP&END */
/* */
/* SET DEFAULT OBJECT CODE */
/* */
/* */
/* NOT END OF RECORD YET */
/* SET STARTING STATE */
/* CLEAR RETURN CODE */
/* POINT TO TOP OF BIT STRUCT */
/* SAVE DEFAULT IN BIT STRUCT */
/* INCREMENT POINTER */
/* SAVE DEFAULT IN BIT STRUCT */
/* INCREMENT POINTER */
/* SAVE DEFAULT IN BIT STRUCT */
/* */
/* POINT IOCB TO INPUT FILE */
/* AND GET NEXT INPUT RECORD */
/* */
/* IF NO PROBLEM - CONTINUE */
/* IF END OF FILE DETECTED- */
/* IF NEITHER- OTHER ERROR */
/* DISPLAY MESSAGE */
/* */
/* AND GO TO MOP UP ROUTINE */
/* CONTINUE */
/* POINT TO START OF RECORD */
/* MOVE SOURCE STATEMENT FOR L */
/* */
/* SAVE CHAR TO CHECK FOR LINE */
/* IF FIRST CHAR A NUMERIC */
/* CHECK NEXT CHARACTER */
/* GET NEXT CHARACTER */
/* IF SPACE, END OF LINE NUMBRE */
/* IF CR FOUND, END OF RECORD */
/* IF NEITHER, CONTINUE TO LO */
/* END OF LINE NUMBER CHECK */
/* IF NO LINE NUMBERS CONTINUE */
/* POINT TO FIRST CHAR AFTER L */
/* PARSE OUT NEXT TOKEN */
/* */
/* PROCESS TOKEN */
/* LOGIC ERROR, SKIP REMAINDER */
/* IF CR AT END, END OF RECORD */
/* SKIP REST OF RECORD */

```

```

3960 GO TO LOOP3
3970 COMMND: IF PARSE1->ONE EQ BUFEND THEN GO TO EORC /* GET NEXT TOKEN */
3980 PARSE1=PARSE1+1 /* MOVE POINTER TO NEXT CHAR */
3990 GO TO COMMND /* REPEAT */
4000 EORC: LOC= /* SET COMMAND PORTION TO BLANK */
4010 INSTL= /*
4020 RTRNT=0 /*
4030 GO TO CONT4 /* MOVE TO PRINT TO OUTPUT */
4040 EORF: IF PARSE1->ONE EQ BUFEND THEN GO TO EOR /*
4050 PARSE1=PARSE1+1 /* SKIP OVER COMMNT AT END OF
4060 GO TO EORF /* RECORD- THEN END OF RECORD
4070 EOR: /* END OF RECORD
4080 PARSE2=ADDR(TOP) /* GET ADDRESS OF BIT STRUCT
4090 NUM=PARSE2->NUM /* GET MS EIGHT BITS OF STRUCT
4100 $ LDA A NUM /* HEX-ASCII OF MS NIBBLE
4110 $ JSR CNVERM /*
4120 $ STA A CHAR6 /* STORE RESULT
4130 $ LDA A NUM /* HEX-ASCII OF LS NIBBLE
4140 $ JSR CNVERL /*
4150 $ STA A CHAR5 /* STORE RESULT
4160 PARSE2=PARSE2+1 /* POINT TO NEXT HEX IN BIT
4170 NUM=PARSE2->NUM /* GET NEXT EIGHT BITS OF STRI
4180 $ LDA A NUM /* HEX-ASCII OF MS NIBBLE
4190 $ JSR CNVERM /*
4200 $ STA A CHAR4 /* STORE RESULT
4210 $ LDA A NUM /* HEX-ASCII OF LS NIBBLE
4220 $ JSR CNVERL /*
4230 $ STA A CHAR3 /* STORE RESULT
4240 PARSE2=PARSE2+1 /* GET NEXT EIGHT BITS OF STRI
4250 NUM=PARSE2->NUM /*
4260 $ LDA A NUM /* HEX-ASCII OF MS NIBBLE
4270 $ JSR CNVERM /*
4280 $ STA A CHAR2 /* STORE RESULT
4290 $ LDA A NUM /* HEX-ASCII OF LS NIBBLE
4300 $ JSR CNVERL /*
4310 $ STA A CHAR1 /* STORE RESULT
4320 INSTL=BREAK /* WRITE HEX-ASCII OBJECT
4330 LOC=INST /* INSTRUCTION COUNT TO OUT
4340 INST=INST+1 /* INCREMENT INSTRUCTION PTR
4350 IF INST EQ 193 THEN DO
4360 BUFFER='ASSEMBLY ABORTED'
4370 CALL DSPLY< , ADDR(BUFFER)>
4380 BUFFER='MORE THAN 192 INSTRUCTIONS'
4390 CALL DSPLY< , ADDR(BUFFER)>
4400 GO TO EOF
4410 END
4420 PARSE2=ADDR(TOP)
4430 BUFFO=PARSE2->BUFFO /* POINT TO TOP OF BIT STRUCT
4440 EPROM=>BUFFO-BUFFO /* LOAD OVER BIT STRING
4450 EPROM=EPROM+3 /* COPY 3 BYTES INTO EPROM
4460 CONT4: LINE=LINE+1 /* INCREMENT POINTER FOR NEXT
4470 ININST=BUFFI /* INCREMENT LINE COUNTER
4480 IF PRINT EQ 1 THEN DO /* MOVE INPUTTED INSTRUCTION
4490 BUFFL=OUT /* IF FILE LISTING REQUESTED
4500 BUFFE(2)=BUFFS(2)+(PARSE1-ADDR(BUFFI))+11 /* MOVE OUT TO RECORD BUFFER
4510 CALL WRITE<1,,> /* SET END POINT
4520 IF RTRN NE 0 THEN GO TO ERR7 /* WRITE LINE
4530 END /* IF I/O PROBLEM-MOPUP&END
4540 IF PRINT EQ 2 THEN CALL DSPLY< , ADDR(OUT)> /*
4550 PAGER: /* IF END OF PAGE
4560 IF LINE GT PAGEL THEN DO /*
4570 CALL TOPPG /* WRITE PAGE HEADER
4580 IF RTRN NE 0 THEN GO TO ERR7 /* IF I/O ERROR-MOPUP&END
4590 END /* END OF PAGE BREAK
4600 IF RTRNT GE 7 THEN DO /* IF ERROR IN ASSEMBLY
4610 LINE=LINE+2 /* INCREMENT LINE COUNTER
4620 RTRNT=0 /* CLEAR STATE
4630 IF PRINT EQ 1 THEN DO /* IF FILE LISTING
4640 BUFFL=MESSGE /* MOVE MESSAGE TO RECORD BUFF
4650 BUFFE(2)=BUFFS(2)+41 /* SET END POINTER
4660 CALL WRITE<1,,> /* WRITE RECORD
4670 IF RTRN NE 0 THEN GO TO ERR7 /* IF I/O PROBLEM-MOPUP&END
4680 END /*
4690 IF PRINT EQ 2 THEN CALL DSPLY< , ADDR(MESSGE)> /* IF PAGE BREAK REQUIRED-GO
4700 IF LINE GT PAGEL THEN GO TO PAGER /* END OF RTRNT GT 8 IF
4710 END /* PICK UP NEXT INPUT RECORD
4720 GO TO LOOP2 /* ALL RECORDS OF INPUT DONE
4730 EOF: /* SAVE TO CHAR NUMBER OF ERRS
4740 NUMPRT=ERROR /* DISPLAY NUMBER OF ERRS
4750 CALL DSPLY< , ADDR(TOTALS)> /* IF FILE LISTING
4760 IF PRINT EQ 1 THEN DO /* MOVE ERRORS MESS TO BUFFL
4770 BUFFL=TOTALS /* SET END OF MESSAGE POINTER
4780 BUFFE(2)=BUFFS(2)+18 /* WRITE MESSAGE TO FILE
4790 CALL WRITE<1,,> /* IF PROBLEMS,MOPUP&END
4800 IF RTRN NE 0 THEN GO TO ERR7
4810 PARSE1=ADDR(RAMNAM)
4820 BUFFE(2)=BUFFS(2)+81
4830 DO INDEX1=1 TO 4
4840 BUFFL=PARSE1->BUFFER
4850 CALL WRITE<1,,>
4860 IF RTRN NE 0 THEN GO TO ERR7
4870 PARSE1=PARSE1+80
4880 END
4890 $ LDX #IOCB+37 /* POINT TO LISTING IOCB
4900 $ SCALL .CLOSE /* CLOSE LISTING FILE
4910 $ IF STATUS(2) NE 0 THEN DO /* IF I/O PROBLEM-MOPUP&END
4920 INDEX1=STATUS(2) /* SAVE ERROR CODE
4930 PRINT=0 /* DON'T TRY CLOSING AGAIN
4940 GO TO ERR7 /* AND GO TO MOPUP
4950 END

```

```

4960 $ LDX #IOCB+37
4970 $ SCALL .RELES
4980 IF STATUS(2) NE 0 THEN DO
4990 INDEX1=STATUS(2)
5000 PRINT=0
5010 GO TO ERR7
5020 END
5030 END
5040 IF OBJECT EQ 1 THEN DO
5050 DO INDEX1=1 TO 8
5060 CALL WRITE<0, >
5070 IF RTRN NE 0 THEN GO TO ERR7
5080 BUFFS(3)=BUFFS(3)+72
5090 BUFFE(3)=BUFFS(3)+71
5100 END
5110 $ LDX #IOCB+74
5120 $ SCALL .CLOSE
5130 IF STATUS(3) NE 0 THEN DO
5140 INDEX1=STATUS(3)
5150 OBJECT=0
5160 GO TO ERR7
5170 END
5180 $ LDX #IOCB+74
5190 $ SCALL .RELES
5200 IF STATUS(3) NE 0 THEN DO
5210 INDEX1=STATUS(3)
5220 OBJECT=0
5230 GO TO ERR7
5240 END
5250 END
5260 $ LDX #IOCB
5270 $ SCALL .CLOSE
5280 IF STATUS(1) NE 0 THEN DO
5290 INDEX1=STATUS(1)
5300 GO TO ERR1
5310 END
5320 $ LDX #IOCB
5330 $ SCALL .RELES
5340 IF STATUS(1) NE 0 THEN DO
5350 INDEX1=STATUS(1)
5360 GO TO ERR1
5370 END
5380 CALL MDOS
5390 END
5400 $CNVERM AND A #0F0H
5410 $ LSR A
5420 $ LSR A
5430 $ LSR A
5440 $ LSR A
5450 $ ADD A #90H
5460 $ DAA
5470 $ ADC A #40H
5480 $ DAA
5490 $ RTS
5500 $CNVERL AND A #0FH
5510 $ ADD A #90H
5520 $ DAA
5530 $ ADC A #40H
5540 $ DAA
5550 $ RTS
5560 PARSE: PROC
5570 LEAD1:
5580 IF PARSE1->ONE EQ ' ' THEN DO
5590 PARSE1=PARSE1+1
5600 GO TO LEAD1
5610 END
5620 TOKENA=' '
5630 INDEX3=0
5640 SCAN: ONE=PARSE1->ONE
5650 IF ONE EQ BUFEND THEN DO
5660 IF INDEX3 EQ 0 THEN DO
5670 RTRNT=1
5680 RETURN
5690 END
5700 RTRNT=2
5710 ENDYET=1
5720 GO TO SEARCH
5730 END
5740 IF ONE EQ '; ' THEN DO
5750 IF INDEX3 EQ 0 THEN DO
5760 RTRNT=5
5770 RETURN
5780 END
5790 RTRNT=4
5800 ENDYET=2
5810 GO TO SEARCH
5820 END
5830 IF ONE EQ ':' THEN DO
5840 RTRNT=3
5850 PARSE1=PARSE1+1
5860 IF INDEX3 EQ 0 THEN DO
5870 RTRNT=7
5880 ERROR=ERROR+1
5890 PD=MESS2
5900 END
5910 RETURN
5920 END
5930 IF ONE EQ ' ' OR ONE EQ ' '
5940 THEN DO
5950 IF INDEX3 EQ 0

```

```

/* POINT TO LISTING IOCB */
/* RELEASE FILE */
/* IF I/O PROBLEM-MOPUP&END */
/* SAVE ERROR CODE */
/* DON'T TRY RELEASING AGAIN */
/* GO TO MOP UP */
/*
/*
/* IF FILE OBJECT
/* WRITE OUT EPROM
/* WRITE OUT RECORD
/* IF I/O PROBLEM-MOPUP&END
/* ADJUST POINTERS FOR NEXT
/* RECORD
/* CONTINUE DO
/* POINT TO OBJECT IOCB
/* CLOSE FILE
/* IF I/O PROBLEM-MOPUP&END
/* SAVE ERROR CODE
/* DON'T TRY TO CLOSE AGAIN
/* GO TO MOPUP
/*
/* POINT TO OBJECT IOCB
/* RELEASE FILE
/* IF PROBLEMS-MOPUP & END
/* SAVE ERROR CODE
/* DON'T TRY RELEASING AGAIN
/* GO TO MOP UP
/*
/* POINT TO INPUT IOCB
/* CLOSE INPUT FILE
/* IF I/O PROBLEM-MOPUP&END
/* SAVE ERROR CODE
/* GO TO MOP UP
/*
/* POINT TO INPUT IOCB
/* AND RELEASE INPUT FILE
/* IF I/O PROBLEM-MOPUP&END
/* SAVE ERROR CODE
/* AND GO TO MOP UP
/*
/* PASS CONTROL BACK TO MDOS
/* END OF MAIN PROC
/* HEX-ASCII OF MS NIBBLE
/* SHIFT MS NIBBLE TO LS NIBBLE
/*
/*
/* RETURN TO MAINLINE
/* HEX-ASCII OF LS NIBBLE
/*
/*
/* RETURN TO MAINLINE
/* TOKEN PARSER ROUTINE
/* REMOVE LEADING SPACES
/* IF A BLANK-IGNORE
/* INCREMENT POINTER
/* AND REPEAT SCAN
/*
/* ASSIGN CURRENT TOKEN BLANK
/* ZERO CURRENT CHARACTER COUN
/* SINGLE OUT CURRENT CHARACTER
/* IF A CR
/* AND TOKEN LENGTH EQ 0
/* SET RETURN CODE, CR NO TOKEN
/* AND RETURN TO MAIN PROGRAM
/*
/* OTHERWISE, SET RETURN CODE
/* DON'T SEARCH RECORD ANY MOR
/* AND SEARCH FOR OPCODE
/*
/* IF A COMMENT DELIMITER
/* AND ZERO LENGTH TOKEN
/* SET RTRNT CODE
/* AND RETURN FOR NEXT RECORD
/*
/* OTHERWISE, SET RETURN CODE
/* PASS OVER REMAINDER OF RECO
/* AND LOOK FOR OPCODE
/*
/* IF LABEL DELIMITER
/* NORMALLY, IGNORE LABEL
/* POINT TO NEXT CHAR FOR PARSE
/* IF ZERO LENGTH, SYNTAX ERROR
/* ZERO LENGTH SYNTAX ERROR
/* INCREMENT ERROR COUNTER
/* SET MESSAGE
/*
/* RETURN TO MAINLINE, NEXT TOK
/*
/* IF NORMAL DELIMITER
/* THEN CHECK FOR ZERO LENGTH
/* IF SO, SYNTAX ERROR

```

```

5960         THEN DO
5970             RTRNT=7
5980             ERROR=ERROR+1
5990             P0=MESS2
6000             RETURN
6010         END
6020         RTRNT=6
6030         PARSE1=PARSE1+1
6040         GO TO SEARCH
6050     END
6060     INDEX3=INDEX3+1
6070     IF INDEX3 GT 8 THEN DO
6080         RTRNT=7
6090         ERROR=ERROR+1
6100         P0=MESS1
6110         RETURN
6120     END
6130     TOKEN(INDEX3)=ONE
6140     PARSE1=PARSE1+1
6150     GO TO SCAN
6160 SEARCH:
6170     IF INDEX3 LE 4 THEN DO
6180         PARSE2=ADDR(TOKENA)
6190         FOUR=PARSE2->FOUR
6200         DO INDEX2=1 TO 87
6210             IF OPCODE(INDEX2) EQ FOUR
6220                 THEN DO
6230                     KIND=CODET(INDEX2)
6240                     RETURN
6250                 END
6260             END
6270         END
6280         PARSE2=ADDR(TOKENA)
6290         KIND=0
6300         NUM=PARSE2->NUM
6310         IF NUM GE 65 AND NUM LE 90 THEN GO TO OK99
6320         ERROR=ERROR+1
6330         P0=MESS16
6340         RTRNT=7
6350         RETURN
6360 OK99:
6370         IF VAR EQ 0 THEN DO
6380             RAMNAM(1)=TOKENA
6390             VAR=1
6400             INDEX2=1
6410             RETURN
6420         END
6430         ELSE DO
6440             DO INDEX2=1 TO VAR
6450                 IF RAMNAM(INDEX2) EQ TOKENA THEN RETURN
6460             END
6470             IF VAR EQ 40 THEN DO
6480                 ERROR=ERROR+1
6490                 P0=MESS14
6500                 RTRNT=7
6510                 RETURN
6520             END
6530             VAR=VAR+1
6540             INDEX2=INDEX2+1
6550             PARSE2=ADDR(RAMNAM)
6560             PARSE2=(PARSE2+(8*VAR))-8
6570             PARSE2->TOKENA=TOKENA
6580             RETURN
6590         END
6600     END
6610 WRITE: PROC(WTYPE, ,)
6620     DCL WTYPE BIN(1)
6630     RTRN=0
6640     IF WTYPE EQ 1
6650         THEN DO
6660             LDX #IOCB+37
6670             SCALL .PUTRC
6680             IF STATUS(2) NE 0
6690                 THEN DO
6700                 BUFFER='LISTING FILE PROBLEM'
6710                 CALL DSPLY< , , ADDR(BUFFER)>
6720                 INDEX1=STATUS(2)
6730                 RTRN=1
6740             END
6750             RETURN
6760         END
6770         ELSE DO
6780             LDX #IOCB+74
6790             SCALL .PUTRC
6800             IF STATUS(3) NE 0
6810                 THEN DO
6820                 BUFFER='OBJECT FILE PROBLEM'
6830                 CALL DSPLY< , , ADDR(BUFFER)>
6840                 INDEX1=STATUS(3)
6850                 RTRN=1
6860             END
6870             RETURN
6880         END
6890     END
6900 OPTION: PROC
6910     RTRN=1
6920     TUNIT=IUNIT
6930     TNAME=INAME
6940     PARSE1=PARSE1+1
6950     PARSE2=PARSE1+1

```

```

/* SET RETURN CODE AND RETURN */
/*
/* INCREMENT ERROR COUNT
/* SET MESSAGE
/* RETURN FOR NEXT TOKEN
/*
/* ELSE, NORMAL RETURN CODE
/* POINT TO NEXT CHAR FOR PARSE
/* AND SEARCH FOR OPCODE
/*
/* IF NONE OF ABOVE, VALID CHA
/* IF TOO MANY CHAR IN TOKEN
/* SET RETURN CODE
/* INCREMENT ERROR COUNT
/* SET MESSAGE
/*
/* ASSIGN VALID CHAR TO TOKEN
/* INCREMENT PARSE POINTER
/* AND CONTINUE SCAN
/*
/* IF POSSIBLE INSTRUCTION
/* POINT TO TOKEN DECODED
/* SAVE FIRST 4 IN 4 LONG STRU
/* SCAN INSTRUCTION TABLE
/* IF TOKEN AND INSTRUCTION
/* THEN SET FLAGS AND RETURN
/* INDEX IN INDEX2
/*
/* END IF
/* CONTINUE DO
/* END LE 4 IF
/*
/* IF NOT INSTRUCTION-VARIABLE
/* GET FIRST CHAR OF TOKEN
/* IF CHAR ALPHA
/* IF FIRST NOT ALPHA, ERROR
/* LOAD ERROR MESSAGE
/* RELAY MESS-ERR
/* AND RETURN TO MAINLINE
/*
/* CHECK IF NO VARS USED YET
/* SAVE TOKEN AS VARIABLE NAME
/* SET RAMNAM POINTER
/* SET POINTER TO FIRST VAR
/* AND RETURN TO MAINLINE
/*
/* ELSE LOOK FOR NAME IN TABLE
/* LOOK THROUGH NAMES IN TABLE
/* INDEX TO RAMNAM IN INDEX2
/* CHECK IF TABLE FULL
/* NEW NAME AND TABLE FULL-ERR
/* INCREM ERR COUNTER&LOAD MES
/* RTRN CODE TO SKIP REST&ERR
/* AND GO BACK TO MAINLINE
/* END VAR EQ 40 IF
/* IF NO ERR, NEW NAME FOR TAB
/* POINT TO VARIABLE
/* CALCULATE ADDRESS FOR TOKEN
/* OFFSET IN RAMNAM
/* STORE TOKENA IN RAMNAM
/* RETURN TO MAINLINE
/* END VAR EQ 0 ELSE DO CLAUSE
/* END OF PARSE PROC
/* LISTING/OBJECT FILE WRITE
/* LISTING OR OBJECT PARM
/* SET NORMAL RETURN CODE
/* IF LISTING FILE WRITE
/* THEN PUT OUT RECORD
/* POINT IOCB TO LISTING FILE
/* WRITE RECORD TO DISKETTE
/* IF PROBLEMS WRITING
/* THEN DISPLAY MESSAGE
/*
/* SAVE RETURN CODE FOR MDERR
/* SET RETURN CODE
/*
/* AND RETURN
/*
/* ELSE OBJECT FILE
/* POINT IOCB TO OBJECT FILE
/* AND WRITE OUT RECORD
/* IF PROBLEM, DISPLAY MESSAGE
/*
/* SAVE RETURN CODE FOR MDERR
/* SET RETURN CODE
/*
/* AND RETURN
/*
/* OPTION PROCESSING ROUTINE
/* SET DEFAULT RETURN CODE
/* SET DEFAULT UNIT NUMBER
/* SET DEFAULT FILE NAME
/* POINT TO NEXT CHAR
/* POINT TO CHAR AFTER
/*

```

```

6960 IF PARSE1->ONE EQ '*' AND PARSE2->ONE EQ '*'
6970 THEN DO
6980 PARSE1=PARSE1+5
6990 RETURN<2, ,>
7000 END
7010 IF PARSE1->ONE EQ '*'
7020 THEN DO
7030 PARSE1->ONE='*'
7040 PARSE2=ADDR(TFILE)
7050 LDX #PARSE1
7060 SCALL .PFNAM
7070 STA B INDEX2
7080 IF INDEX2 LE 1
7090 THEN DO
7100 PARSE1=PARSE1+1
7110 GO TO OK0
7120 END
7130 ELSE DO
7140 RTRN=2
7150 RETURN<0, ,>
7160 END
7170 END
7180 OK0: FILE(INDEX3)=TFILE
7190 LDX #IOCB
7200 STX PARSE2
7210 PARSE2=PARSE2+OFFSET
7220 LDX PARSE2
7230 SCALL .RESRV
7240 IF STATUS(INDEX3) NE 0
7250 THEN DO
7260 INDEX1=STATUS(INDEX3)
7270 RTRN=3
7280 RETURN<0, ,>
7290 END
7300 LDX PARSE2
7310 SCALL .OPEN
7320 IF STATUS(INDEX3) NE 0
7330 THEN DO
7340 INDEX1=STATUS(INDEX3)
7350 LDX PARSE2
7360 SCALL .RELES
7370 RTRN=3
7380 RETURN<0, ,>
7390 END
7400 RETURN<1, ,>
7410 END
7420 TOPPG: PROC
7430 CPT4=PAGE
7440 PAGE=PAGE+1
7450 LINE=3
7460 RTRN=0
7470 IF PRINT EQ 1 THEN DO
7480 BUFL=HEADER
7490 BUFE(2)=BUFS(2)+79
7500 CALL WRITE<1, ,>
7510 IF RTRN NE 0 THEN GO TO ENDTOP
7520 BUFL=HEAD2
7530 BUFE(2)=BUFS(2)+29
7540 CALL WRITE<1, ,>
7550 IF RTRN NE 0 THEN GO TO ENDTOP
7560 END
7570 IF PRINT EQ 2 THEN DO
7580 CALL DSPLY<, , ADDR(BUFL)>
7590 CALL DSPLY<, , ADDR(HEAD2)>
7600 BUFFER=' '
7610 CALL DSPLY<, , ADDR(BUFFER)>
7620 END
7630 ENDTOP:
7640 RETURN
7650 END
7660 PROCES: PROC
7670 PARSE2=ADDR(BITUP)
7680 IF KIND EQ 0 THEN NUM=INDEX2-1
7690 ELSE NUM=CODE(INDEX2)
7700 PARSE2->NUM=NUM
7710 IF KIND EQ 6 THEN GO TO NONSUP
7720 GO TO (ST1,ST2,ST3,ST4,ST5,ST6,STAERR),STATE
7730 ST1: IF RTRNT EQ 3 THEN DO
7740 STATE=2
7750 GO TO ENDSTA
7760 END
7770 IF KIND EQ 1 THEN GO TO ARITH
7780 IF KIND EQ 5 THEN GO TO IOCODE
7790 GO TO STAERR
7800 ST2: IF KIND EQ 1 THEN GO TO ARITH
7810 IF KIND EQ 5 THEN GO TO IOCODE
7820 GO TO STAERR
7830 ST3: IF KIND EQ 0 THEN GO TO VARDES
7840 IF KIND EQ 2 THEN GO TO DARDES
7850 GO TO STAERR
7860 ST4: IF KIND EQ 0 THEN GO TO VARSOU
7870 IF KIND EQ 2 THEN GO TO DARSOU
7880 IF KIND EQ 3 THEN GO TO CONXXX
7890 GO TO STAERR
7900 ST5: IF KIND EQ 5 THEN GO TO IOCODE
7910 IF KIND EQ 4 THEN GO TO SHXXX
7920 GO TO STAERR
7930 ST6: IF KIND EQ 5 THEN GO TO IOCODE
7940 STAERR: ERROR=ERROR+1
7950 P0=MESS2

```



```

PNT90: STATE=8                                /* INDICATE ERROR TO MAINLINE */
7970 RTRNT=7                                    /* */
7980 RETURN                                      /* */
7990 ARITH: STATE=3                             /* RETURN TO MAINLINE */
8000 IF INDEX2 EQ 2 THEN ANDX=1                /* SET NEW STATE CODE */
8010 IF INDEX2 EQ 3 THEN LIMX=1
8020 IF INDEX2 EQ 4 THEN ABSX=1
8030 IF INDEX2 EQ 6 THEN SUBX=1
8040 L0=0
8050 L1=0
8060 L2=0
8070 IF BIT0 EQ 1 THEN L0=1
8080 IF BIT1 EQ 1 THEN L1=1
8090 IF BIT2 EQ 1 THEN L2=1
8100 GO TO ENDSTA
8110 ICODE: STATE=7                             /* RETURN TO MAINLINE */
8120 IF INDEX2 EQ 16 THEN GO TO PNT97          /* SET NEW STATE CODE */
8130 IF INDEX2 GE 33 AND INDEX2 LE 40 THEN DO
8140 PNT97: IF SUBX NE 1 THEN GO TO PNT98
8150 GO TO PNT96
8160 END
8170 GO TO PNT99
8180 PNT96: CONSUB(1)=1
8190 IF DAROUT(1) EQ 1 THEN DO
8200 ERROR=ERROR+1
8210 P0=MESS12
8220 GO TO PNT90
8230 END
8240 GO TO PNT99
8250 PNT98: IF ANDX EQ 1 THEN DO
8260 ERROR=ERROR+1
8270 P0=MESS9
8280 GO TO PNT90
8290 END
8300 IF LIMX EQ 1 THEN DO
8310 ERROR=ERROR+1
8320 P0=MESS10
8330 GO TO PNT90
8340 END
8350 IF ABSX EQ 1 THEN DO
8360 ERROR=ERROR+1
8370 P0=MESS11
8380 GO TO PNT90
8390 END
8400 PNT99: IF INDEX2 EQ 15 THEN GO TO PNT100
8410 IF INDEX2 GE 25 AND INDEX2 LE 32 THEN DO
8420 PNT100: CVTX(1)=1
8430 IF DAROUT(2) EQ 1 THEN DO
8440 ERROR=ERROR+1
8450 P0=MESS5
8460 GO TO PNT90
8470 END
8480 IF CVTX(2) EQ 1 THEN DO
8490 ERROR=ERROR+1
8500 P0=MESS6
8510 GO TO PNT90
8520 END
8530 IF CONSUB(2) EQ 1 THEN DO
8540 ERROR=ERROR+1
8550 P0=MESS7
8560 GO TO PNT90
8570 END
8580 IF DAROUT(1) EQ 1 THEN DO
8590 ERROR=ERROR+1
8600 P0=MESS8
8610 GO TO PNT90
8620 END
8630 END
8640 IF INDEX2 GE 17 AND INDEX2 LE 24 THEN DO
8650 IF DAROUT(2) EQ 1 THEN DO
8660 ERROR=ERROR+1
8670 P0=MESS3
8680 GO TO PNT90
8690 END
8700 IF CONSUB(2) EQ 1 THEN DO
8710 ERROR=ERROR+1
8720 P0=MESS4
8730 GO TO PNT90
8740 END
8750 END
8760 ADF0=0
8770 ADF1=0
8780 ADK0=0
8790 ADK1=0
8800 ADK2=0
8810 IF BIT0 EQ 1 THEN ADF0=1
8820 IF BIT1 EQ 1 THEN ADF1=1
8830 IF BIT2 EQ 1 THEN ADK0=1
8840 IF BIT3 EQ 1 THEN ADK1=1
8850 IF BIT4 EQ 1 THEN ADK2=1
8860 GO TO ENDSTA
8870 DARDES:
8880 DAROUT(1)=1
8890 VARDES: STATE=4
8900 B0=0
8910 B1=0
8920 B2=0
8930 B3=0
8940 B4=0
8950 B5=0

```

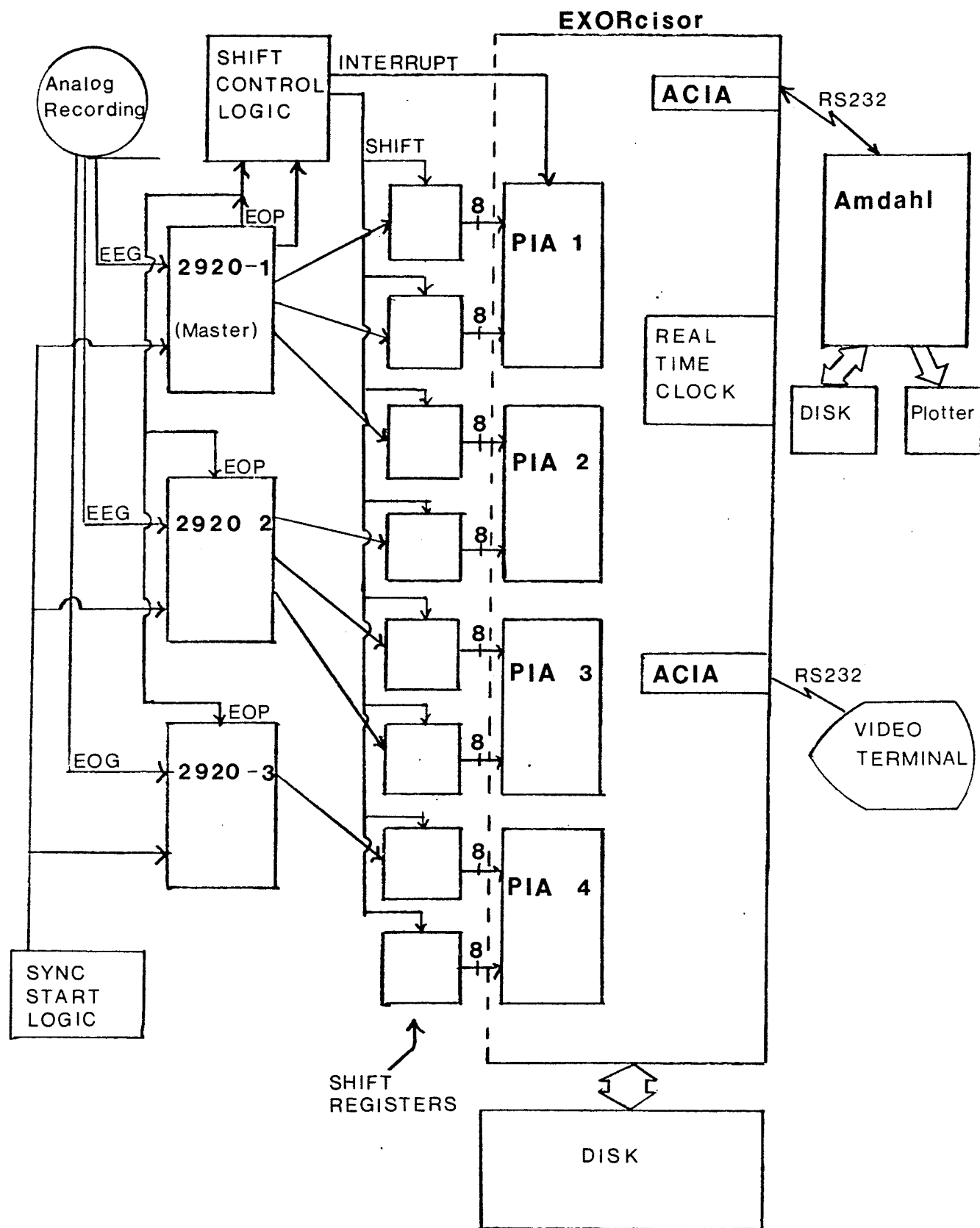
```

8960 IF BIT0 EQ 1 THEN B0=1
8970 IF BIT1 EQ 1 THEN B1=1
8980 IF BIT2 EQ 1 THEN B2=1
8990 IF BIT3 EQ 1 THEN B3=1
9000 IF BIT4 EQ 1 THEN B4=1
9010 IF BIT5 EQ 1 THEN B5=1
9020 GO TO ENDSTA
9030 DARSOU:
9040 VARSOU:
9050 CONY STATE=5
9060 A0=0
9070 A1=0
9080 A2=0
9090 A3=0
9100 A4=0
9110 A5=0
9120 IF BIT0 EQ 1 THEN A0=1
9130 IF BIT1 EQ 1 THEN A1=1
9140 IF BIT2 EQ 1 THEN A2=1
9150 IF BIT3 EQ 1 THEN A3=1
9160 IF BIT4 EQ 1 THEN A4=1
9170 IF BIT5 EQ 1 THEN A5=1
9180 GO TO ENDSTA
9190 SHXXX: STATE=6
9200 IF LIMX EQ 1 AND CODE(INDEX2) NE $OF THEN DO
9210 ERROR=ERROR+1
9220 P0=MESS13
9230 GO TO PNT90
9240 END
9250 S0=0
9260 S1=0
9270 S2=0
9280 S3=0
9290 IF BIT0 EQ 1 THEN S0=1
9300 IF BIT1 EQ 1 THEN S1=1
9310 IF BIT2 EQ 1 THEN S2=1
9320 IF BIT3 EQ 1 THEN S3=1
9330 GO TO ENDSTA
9340 NONSUP: ERROR=ERROR+1
9350 P0=MESS15
9360 STATE=8
9370 RTRNT=7
9380 RETURN
9390 ENDSTA: RETURN
9400 END

/* RETURN TO MAINLINE */
/*
/*
/* SET NEW STATE CODE */
/* COPY IN BIT ADDRESS */
/*
/*
/*
/*
/*
/*
/* RETURN TO MAINLINE */
/* SET NEW STATE CODE */
/*
/*
/* COPY IN SHIFT BIT CODE */
/*
/*
/*
/*
/* NON-SUPPORTED INSTRUCTION */
/*
/* FLAG PROBLEM TO MAINLINE */
/*
/* RETURN TO MAINLINE */
/*

```

Appendix C
CIRCUIT DIAGRAMS



SYSTEM CONFIGURATION

