

MICROPROCESSOR BASED SPECTRUM ANALYSER

by KEVIN J ROGERS B E (Honours)

being submitted as a thesis for the  
degree of MASTER OF ENGINEERING SCIENCE  
at the University of Adelaide, October 1979.

This thesis embodies the results of  
supervised project work making up  
two-thirds of the work for the degree.

## SUMMARY

The purpose of this project is to make a low frequency spectrum analyser which has good resolution and dynamic range but is not too expensive. This is achieved by digitizing the analogue input signal and calculating the spectrum using an 8 bit INTELL 8080A micro-processor.

The method for digitally calculating the spectrum of signals has been exploited considerably since Cooley and Tukey introduced the Fast Fourier Transform in the 1960's and the fourier theory used in this project has already been well developed over the past twenty years. The uniqueness of this project lies in the fact that an 8-bit microprocessor is used for a complex "number crunching" application formerly reserved for larger and more powerful computers or minicomputers. Thus a flexible instrument with considerable potential has been built for a capital cost of approximately \$1 000 which is much less than commercial units currently available. However the disadvantage of using a microprocessor is the programming time required to generate efficient software.

The hardware for the spectrum analyser consists of the micro-processor system, power supply, analogue to digital and digital to analogue circuitry and input filters. It was also found necessary to build a digital hardware multiplier to keep calculation time to a reasonable level. The analyser is used in conjunction with a Cathode Ray Oscilloscope or a paper recorder to provide a medium for observing results.

The software was written in INTELL 8080 assembly language. The reason for choosing assembly language rather than a high level language such as PLM was to minimise required memory capacity and execution time.

The spectrum analyser was built with a view to analysing mechanical vibrations at Torrens Island Power Station. The use for which the instrument was built determined the specification and hence the design approach but the application method is not the subject of this thesis.

### DECLARATION

This thesis contains no material which has been accepted for the award of any other degree or diploma in any university and to the best of the author's knowledge and belief contains no material previously published or written by another person, except where due reference is made in the text of the thesis.

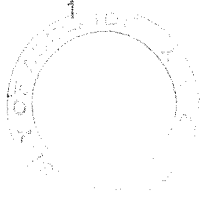
### ACKNOWLEDGEMENTS

Thanks are due to my supervisors Dr B Davis and Dr D Pucknell from the University of Adelaide and to a technical officer at the Torrens Island Power Station, R Vear, for his assistance in construction and debugging.

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The recent advances in large-scale integrated circuit technology have brought substantial improvements to the performance of instrumentation and control equipment. In particular, the micro-processor has opened up the possibility of new approaches to portable equipment which, for reasons of cost and size, were formerly impracticable. In this project Fourier Transform techniques in the form of the Discrete Fourier Transform are applied using a micro-processor as the calculating medium to produce a Spectrum Analyser.

#### 1.1 Alternative Approaches to Spectrum Analysis

The purpose of a spectrum analyser is basically to display the frequency content of a signal giving also the magnitude of the frequency components that are present. This subject has been of interest to communication and acoustic engineers for many years and, of late, there has been a fast growing use of spectrum analysis for the examination of mechanical vibration. There are different methods of achieving spectrum analysis but in the following paragraphs three fundamental approaches are listed :

##### 1.1 (a) Third Octave Filters

In this approach the input signal is fed to the inputs of a bank of bandpass filters. The outputs of the filters are sequentially selected by a switching network and fed to a unit which measures the magnitude of the output and this is fed in turn to a recording instrument. Typically the centre frequencies are separated by  $1/3$ rd of an octave. Hence each filter has a passband of approximately 50% of the centre frequency.



The resolution is poor and for low frequency work it is time consuming but this type of analyser has still been used for many applications, especially acoustics.

1.1 (b) Spectrum Analyser Based on Heterodyne Principle

This analyser operates on the principle that the input signal is multiplied with a local oscillator LO. The result is a signal consisting of sum and difference frequencies which are then fed to a fixed frequency bandpass filter. By sweeping the frequency of the local oscillator the magnitude of a range of frequency components can be detected on the output of the bandpass filter. This approach has a constant bandwidth resolution over the frequency range of interest and is the predominant technique used for analysing communication frequencies. However at low frequencies (where the bandwidth of the filter is required to be small) this approach is time consuming to operate.

1.1 (c) Digital Spectrum Analysis

This is achieved by sampling the input signal and then calculating the spectrum from the samples using a digital computing unit. The basic calculation method is called the Digital Fourier Transform (DFT) and is generally implemented by a faster calculation routine called the Fast Fourier Transform (FFT). More detail is given regarding these methods in later chapters.

This approach is far superior for low frequencies as the operation time is virtually confined to the time taken to sample the input signal. The upper frequency limit is of the

order of 1 MHz, the limit being due to maximum sampling rates of analogue to digital converters (A/D) and computer processing speeds.

Digital Spectrum analysis has been implemented by recording the signal on location and then 'playing back' to an 'in house' computer or alternatively transporting a mini-computer to the location to get direct results. Units employing the latter method have been available for the last couple of years for prices in the range : \$20 000 - \$50 000.

## 1.2 The Application

This project arose out of the need for a spectrum analyser to examine mechanical vibrations at the Torrens Island Power Station. Typical applications are examination of turbo - generator rotor vibrations to assist in a preventative maintenance program or to predict the cause of a particular vibration mode. The analyser would be useful to monitor most other rotating machinery at the power station and also to examine boiler vibrations.

The use of an analyser for these purposes is a complicated study in itself and is not covered in this thesis but the application for which the analyser was designed determined the specifications and the approach required. These are :

- (a) Highest frequency of interest is the order of 5 KHz.
- (b) Linear Frequency response since detection of harmonics of fundamentals is required.
- (c) High resolution.

These requirements can be most easily met using digital spectrum analysis. However, for many applications the price tag of \$20 000 to \$50 000 for a minicomputer model can be prohibitive. It was hence decided to attempt to build a spectrum analyser which met the requirements of the application by using a microprocessor as the computational element. This proposition has several advantageous features :

- (a) Capital cost of parts is relatively cheap (of the order of \$1 000) thus presenting the possibility of using the spectrum analyser for dedicated applications.
- (b) Microprocessor software can be modified easily to suit the particular user's application.
- (c) The user could use the same instrument for entirely different applications simply by exchanging software eg. it could be used for a general purpose input-output controller.

1.3 Specifications for the Spectrum Analyser

The specifications for the spectrum analyser were determined such that maximum use could be made of the vibrometers with which it would be used. They are as follows :

- (a) Input Voltage Range :

Maximum peak to peak input voltage is selectable in 10 db steps from 10 mV p-p to 1 V p-p.

- (b) Frequency Range :

- 0 - 100 Hz
- 0 - 200 Hz
- 0 - 500 Hz

- 0 - 1 KHz
- 0 - 2 KHz
- 0 - 5 KHz

ie. there are 6 frequency ranges which are selectable and there are 100 frequency output points in each range. Therefore the maximum frequency resolution is nominally 1%.

(c) Dynamic Range :

A dynamic range of 50 db is adequate.

2.1 The Fourier Transform

For an input signal  $x(t)$  the fourier transform (FT) is defined as :

$$X(f) = \int_{-\infty}^{+\infty} x(t) \exp(-j 2\pi ft) dt$$

and gives the magnitude of the frequency content of  $x(t)$  at any frequency  $f$ .

The inverse fourier transform (IFT) can be shown to be :

$$x(t) = \int_{-\infty}^{+\infty} X(f) \exp(j 2\pi ft) df$$

2.2 Properties of the Fourier Transform

1. If  $x(t)$  is real then

$$X(-f) = X^*(f)$$

2. Let  $X(f)$ ,  $Y(f)$ ,  $Z(f)$  be the fourier transforms of  $x(t)$ ,  $y(t)$ ,  $z(t)$  respectively.

$$\text{if } z(t) = \int_{-\infty}^{+\infty} x(\lambda) y(t - \lambda) d\lambda$$

$$\text{then } Z(f) = X(f) \cdot Y(f)$$

$$\text{if } Z(f) = \int_{-\infty}^{+\infty} X(\lambda) Y(f - \lambda) d\lambda$$

$$\text{then } z(t) = x(t) \cdot y(t)$$

ie. convolution in the time domain corresponds to multiplication in the frequency domain and vice versa.

2.3 Practical Limitations

When the fourier integral is evaluated numerically there are three limitations :

1. The signal  $x(t)$  is evaluated over a finite interval.
2. The signal is sampled at a finite number of points within that interval.

3. The transform is calculated for a finite number of frequencies.

The first two of these limitations are a source of error and their effects and remedies need to be considered :

$$\begin{aligned}\text{Let } \bar{X}(f) &= \int_{-T/2}^{T/2} x(t) \exp(-j 2\pi ft) dt \\ &= \int_{-\infty}^{+\infty} \text{rect}(t/T) x(t) \exp(-j 2\pi ft) dt\end{aligned}$$

$$\text{Where } \text{rect}(t) = \begin{cases} 1 & \text{for } -0.5 \leq t \leq 0.5 \\ 0 & \text{otherwise} \end{cases}$$

$$\text{The F.T. of } \text{rect}(t/T) = \text{sinc}(f)$$

$$\text{Where } \text{sinc}(f) = \begin{cases} 1 & \text{for } f = 0 \\ \frac{\sin \pi f}{\pi f} & \text{otherwise} \end{cases}$$

By applying the rule that multiplication in the time domain corresponds to convolution in the frequency domain it is obvious that :

$$\bar{X}(f) = \int_{-\infty}^{+\infty} X(\lambda) T \text{sinc}(f - \lambda) T d\lambda$$

In other words the true spectrum is 'smeared' by the sinc function, the effect of which can be illustrated by the following example for a 5Hz sinusoid :

$$\text{Let } x(t) = \cos 2\pi(5)t$$

$$\therefore X(f) = \frac{1}{2} \delta(f-5) + \frac{1}{2} \delta(f+5)$$

Where  $\delta$  is the dirac - delta function. Figures 2.1 and 2.2 show  $x(t)$  and  $X(f)$  respectively.

If  $x(t)$  is measured only over the interval  $(-0.5, 0.5)$  seconds then the estimate for  $X(f)$  would be :

$$\bar{X}(f) = \frac{1}{2} \text{sinc}(f-5) + \frac{1}{2} \text{sinc}(f+5)$$

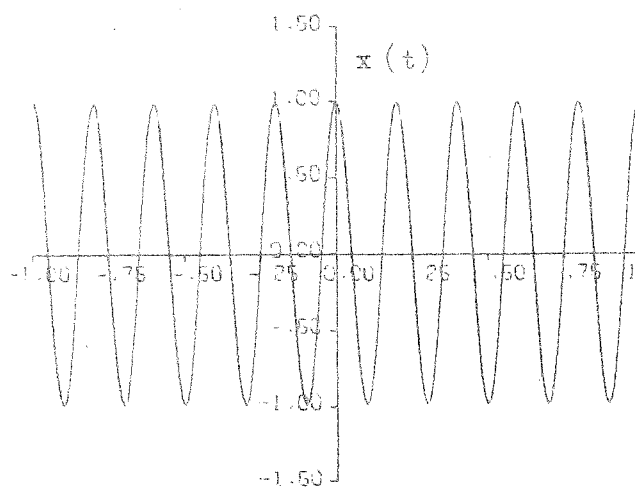


Figure 2.1 :  $x(t)$

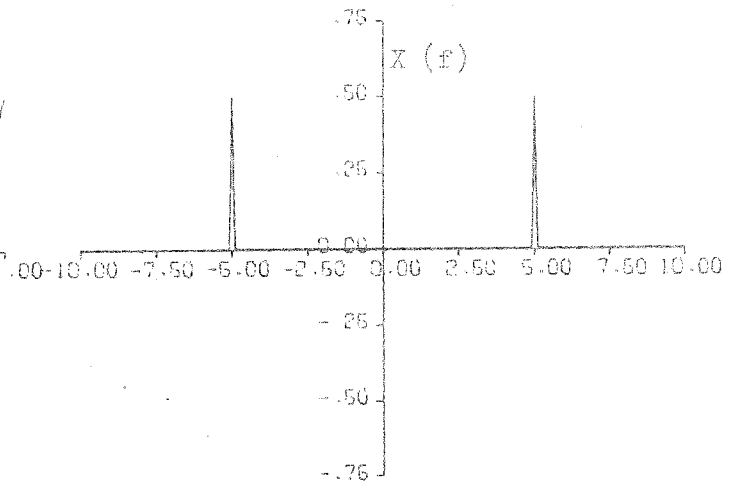


Figure 2.2 :  $X(f)$

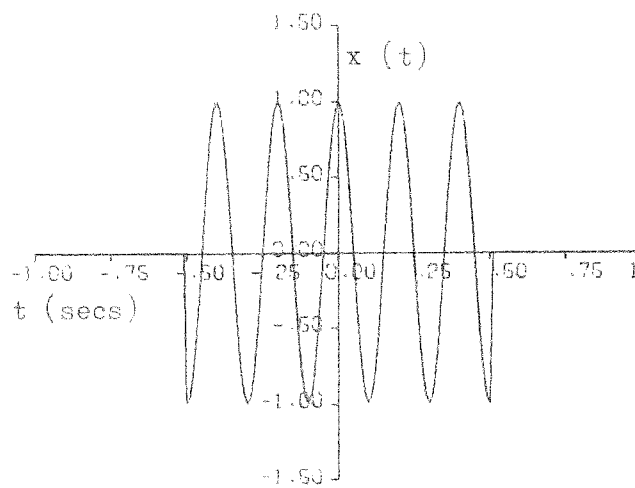


Figure 2.3 :  $\bar{x}(t)$

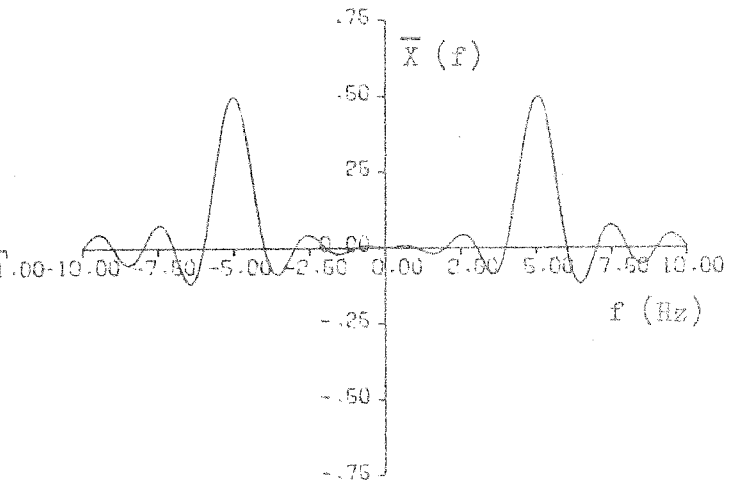


Figure 2.4 :  $\bar{X}(f)$

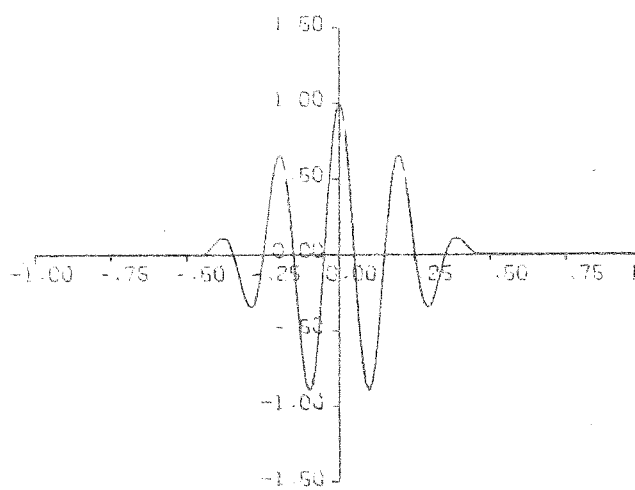


Figure 2.5 :  $\tilde{x}(t)$

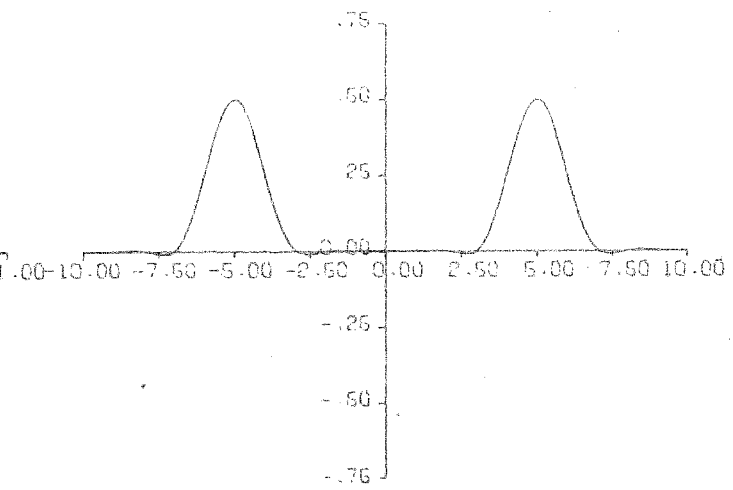


Figure 2.6 :  $\tilde{X}(f)$

Figures 2.3 and 2.4 show  $\bar{x}(t)$  and  $\bar{X}(f)$  respectively and it is obvious that the original spectrum has been degraded by the appearance of sidelobes which may possibly conceal another smaller frequency component at say 7.5 Hz.

The problem can be partially overcome by premultiplying the input data by a window function  $w(t)$ . This function is chosen to minimise the sidelobes in the frequency domain and thus increase the selectivity of the analyser. As an example the half raised cosine will be considered.

$$\text{ie. } w(t) = \begin{cases} \frac{1}{2} + \frac{1}{2} \cos 2\pi t & \text{for } -0.5 \leq t \leq 0.5 \\ 0 & \text{otherwise} \end{cases}$$

$$W(f) = 0.25 \text{ sinc}(f-1) + 0.5 \text{ sinc}(f) + 0.25 \text{ sinc}(f+1)$$

$$\text{Let } \tilde{x}(t) = x(t) \cdot w(t)$$

$$\text{Let } \tilde{X}(f) = \text{FT of } \tilde{x}(t)$$

These two functions are shown in figures 2.5 and 2.6 respectively.

The choice of the window function is a subject of debate. The windows chosen for this project were the half raised cosine window and a modified Bingham Window. These are shown in figures 2.9 to 2.12 together with the magnitude of the spectrum (expressed in db). Figures 2.7 and 2.8 show the rectangular function and its transform for the sake of comparison.

The second limitation leads to the problem of aliasing distortion and can be expressed in the following manner.

Let  $x(t)$  be sampled at rate  $f_c$  times/second.

∴ the sampled signal:

\* "Modified Bingham" window is the name used in the remainder of the text to refer to the window function shown in figure 2.11 which is similar to that proposed by Bingham et al in reference [4].



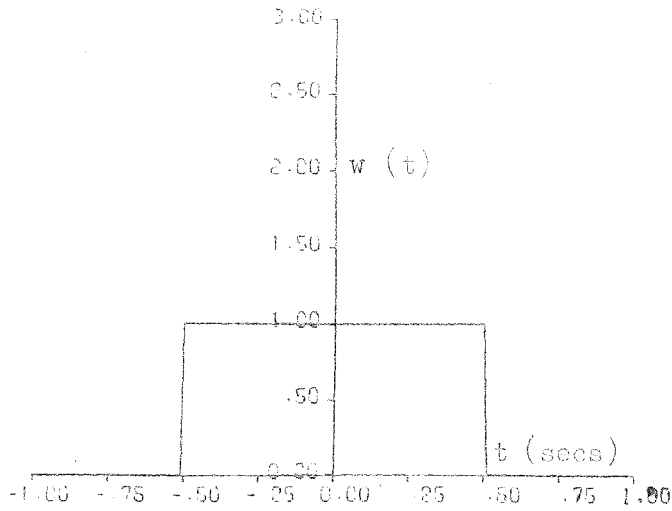


Figure 2.7 : Rectangular Window

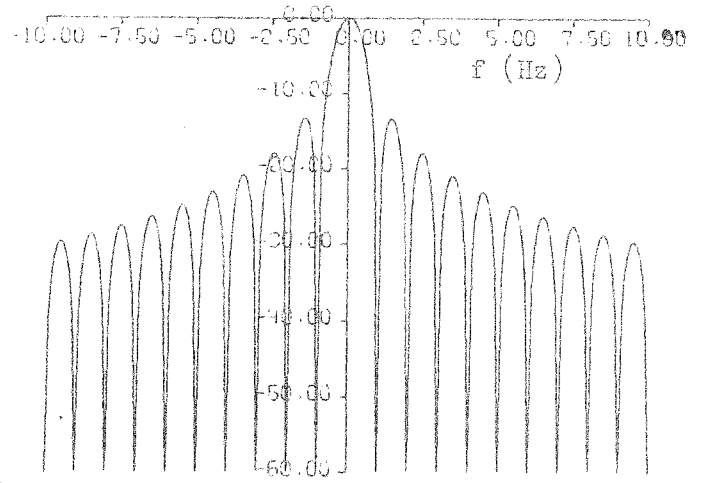


Figure 2.8 :

Spectrum of Rectangular Window

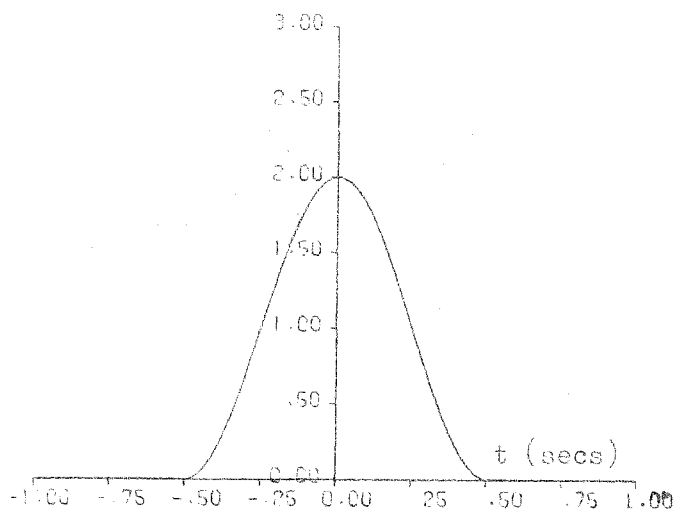


Figure 2.9 : Hanning Window

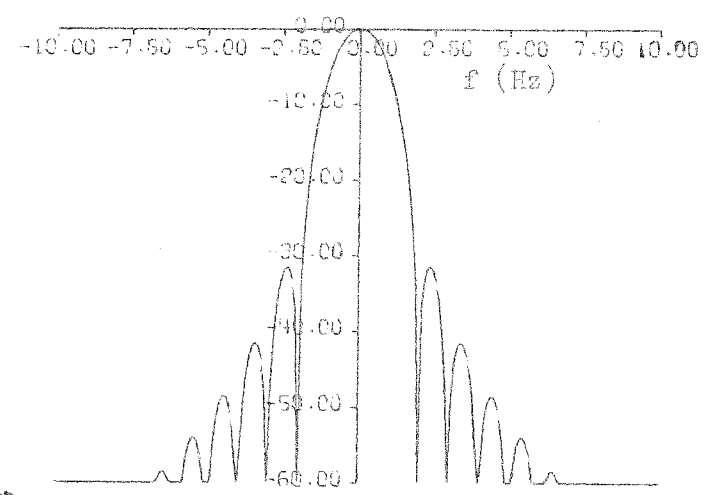


Figure 2.10 : Hanning Window Spectrum

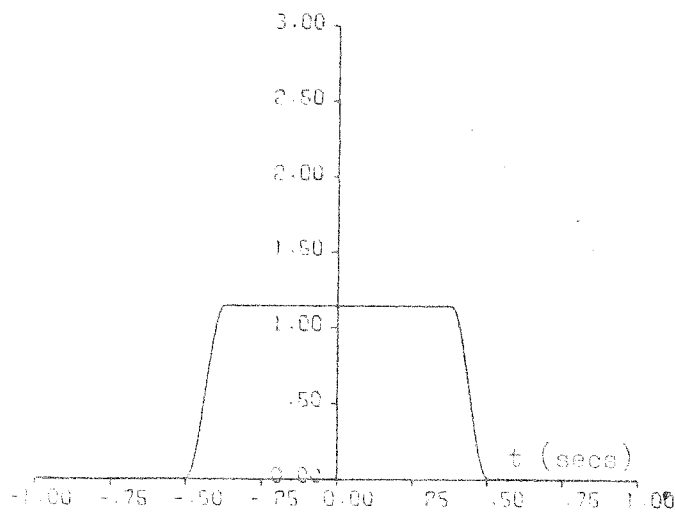


Figure 2.11 : Modified Bingham

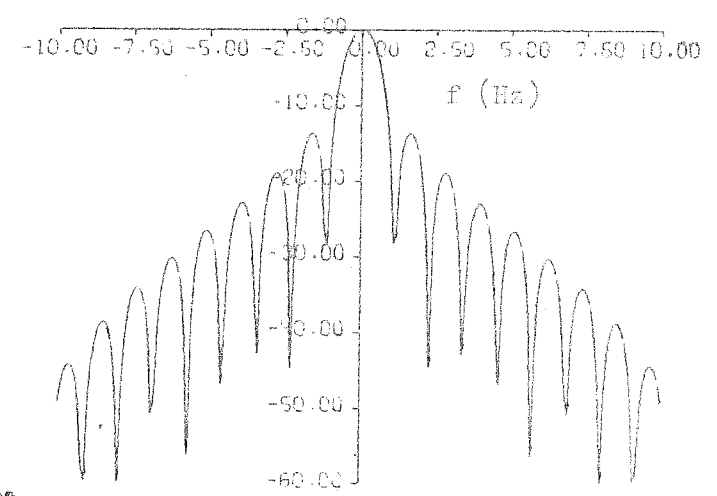


Figure 2.12 :

Spectrum of Modified Bingham

$$\hat{x}(t) = x(t) \sum_{n=-\infty}^{+\infty} \delta(t - n\Delta t)$$

$$\text{Where } \Delta t = 1/f_c$$

$$\therefore \hat{X}(f) = \sum_{n=-\infty}^{+\infty} X(f - fc)$$

From this equation it can be shown that  $X(f)$  is periodic with period  $fc$  and that if  $X(f)$  has components greater than  $fc/2$  the adjacent spectra interfere with each other. However, if the input signal is filtered to eliminate components greater than  $fc/2$  then:

$$\hat{X}(f) = X(f) \text{ for } |f| < fc/2.$$

#### 2.4 Numerical Evaluation of the Fourier Transform

$$\text{Let } \bar{X}(f) = \int_0^T x(t) \exp(-j 2\pi ft) dt$$

By applying the trapezoidal rule of integration this can be approximated by :

$$\bar{X}(f) \approx \left\{ \sum_{i=0}^{N-1} x(i\Delta t) \exp(-j 2\pi fi\Delta t) - x(0) + x(N\Delta t) \exp(-j 2\pi fN\Delta t) \right\} \Delta t$$

$$\text{Where } \Delta t = T/N$$

$\bar{X}(f)$  is only calculated at a finite number of points  $\therefore$

$$\text{if } \Delta f = 1/T$$

$$\bar{X}(k\Delta f) = \Delta t \left\{ \sum_{i=0}^{N-1} x(i\Delta t) \exp(-j 2\pi ik/N) - x(0) + x(T) \right\}$$

$$\text{Let } x_i = x(i\Delta t)$$

$$\bar{X}_k = \bar{X}(k\Delta f)$$

$$\therefore \bar{X}_k = \Delta t \left\{ \sum_{i=0}^{N-1} x_i \exp(-j 2\pi ik/N) - x_0 + x_N \right\}$$

The expression

$$X_k = \sum_{i=0}^{N-1} x_i \exp(-j 2\pi ik/N)$$

is an approximation to the Fourier Transform and is known as the Discrete Fourier Transform (DFT). If this expression is evaluated directly for  $N$  frequencies then  $N^2$  complex multiplications need to be executed which can require a large amount of computer time as  $N$  becomes large. However in 1965 Cooley and Tukey [1] published an algorithm for a more efficient method to evaluate the DFT which is called the Fast Fourier Transform (FFT). This technique is well documented and will not be outlined here but the particular method used for this project is the decimation in time routine outlined in pages 173 - 182 of reference [2]. Where  $N$  is a power of 2 the number of complex multiplications required is now  $\frac{1}{2}N \log_2 N$ . For this project a 256 point transform is used. Hence calculation saving ratio =  $(N^2 / (\frac{1}{2}N \log_2 N)) = 2N / \log_2 N$   
 = 64 times for  $N = 256$ .

### 2.2 Power Spectral Density (PSD)

The power spectral density of signal  $x(t)$  at frequency  $f$  is defined as :

$$G(f) = \lim_{\Delta f \rightarrow 0} \frac{1}{\Delta f} \left\{ \lim_{T \rightarrow \infty} \frac{1}{T} \int_{(f - \Delta f/2)}^{(f + \Delta f/2)} |X(\lambda)|^2 d\lambda \right\}$$

i.e. the average power between frequencies  $f_1$  and  $f_2 =$

$$\int_{f_1}^{f_2} G(f) df.$$

Let  $S_k = G(k\Delta f)$

It can be shown [3] that  $S_k \approx |X_k|^2$

If the input signal has a random component as well as a fixed frequency component then the spectrum measured from one transform to the next will differ. However if many spectra are calculated and then averaged the error is reduced in proportion to the square root of the number of spectra averaged.

eg. this analyser can average up to 128 spectra which gives an increase in accuracy of measurement greater than 11 times compared with a single calculation.

References:

- [1] "An algorithm for the machine calculation of complex Fourier series" by J Cooley and J Tukey. Math. Comput., 19, p.297, 1965.
- [2] "Digital Processing of Signals" by Gold and Rader.
- [3] "Random Vibrations and Spectral Analysis" by D. E. Newland.
- [4] "Modern Techniques of Power Spectrum Estimation" by C. Bingham, M.D. Godfrey and J.W. Tukey, IEEE Trans. Audio and Electroacoustics, Vol. AU-15, 1967, No.2, 56-66.

HARDWARE

The controlling and calculating component of the spectrum analyser is an INTELL 8080A microprocessor which is coupled to inputs and outputs via 8255 input-output ports. The structure used is similar to that of the SDK-80 kit but the CPU, ROM and an extended RAM are built on separate circuit boards using wire-wrap for construction. For the sake of simplicity, the diagrams used in this chapter will be basically block diagram form but detailed circuit diagrams are given in Appendix 4.

## 3.1

THE CENTRAL PROCESSOR UNIT (figure 3.1)

The 8080 microprocessor integrated circuit is used in conjunction with two other IC's which are the 8224 clock generator and the 8228 System controller. The clock generator uses an 18.432 Mhz crystal and divides this by 9 to produce a 2.048 Mhz 2 phase clock for the 8080. An asynchronous reset signal is received by the 8224 and clocked into the 8080 at the correct machine cycle. In this way the microprocessor program counter is reset to zero and program execution is initiated from the first location in memory. The 8228 system controller reads an 8 bit data stream plus some other control signals from the 8080 and decodes these into the 8 bit bidirectional data bus and the control bus. This 3 chip arrangement is standard for the 8080 system but later devices substitute 1 integrated circuit to achieve the same effect (eg the 8085). The two 8212 input-output ports serve as an address bus buffer which is required to drive the large memory array. An 7494 flip-flop is used to divide the 18.432 Mhz by 2 to provide a 9.216 Mhz clock for the hardware multiplier and 3 counters divide the 18.432 Mhz by 2640 to provide a 6982 Hz clock for the teletype board.

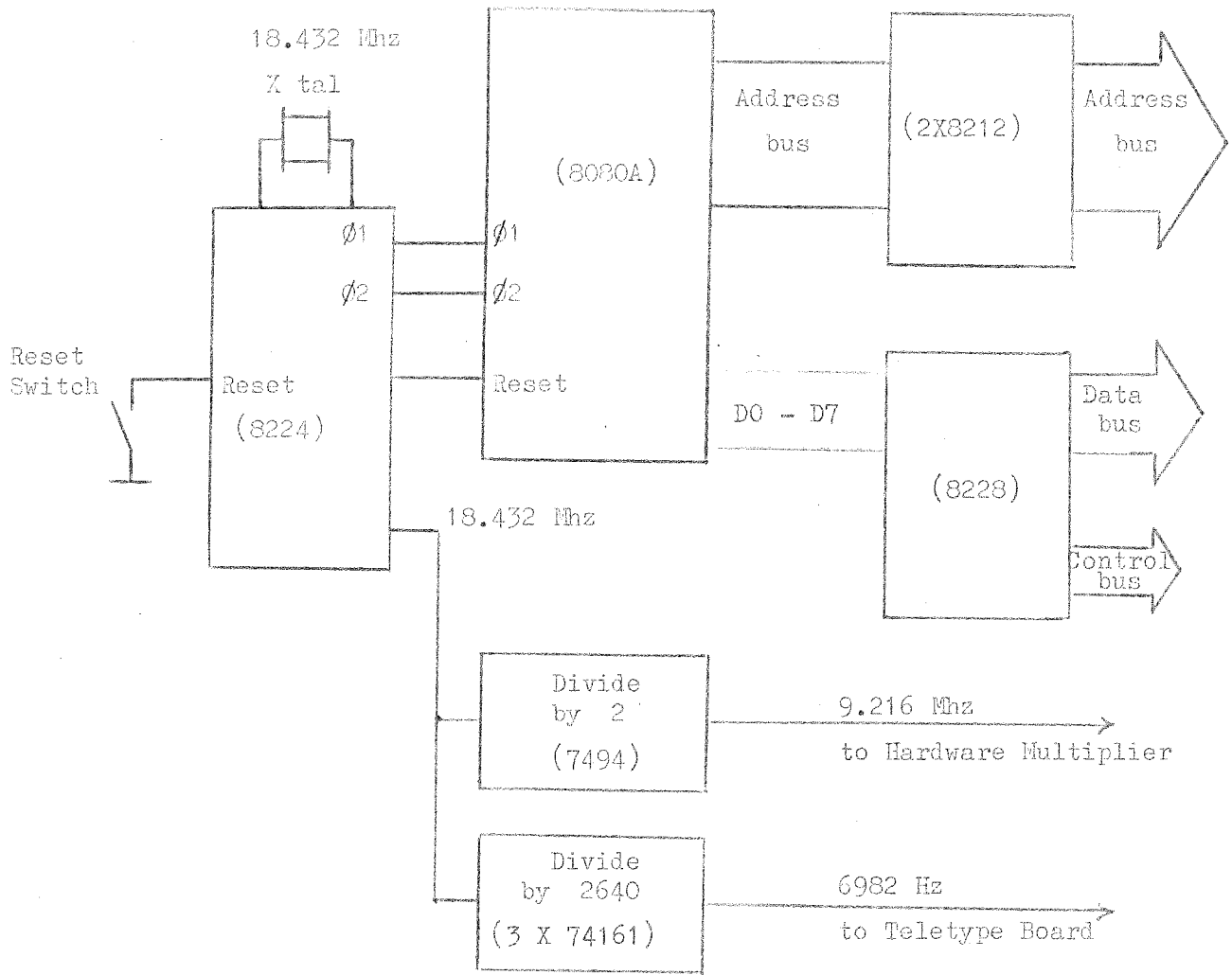


Figure 3.1 : CPU Board

3.2

READ ONLY MEMORY

A circuit board is reserved for read only memory (ROM) in which program code and look up tables are stored. The block diagram for the circuit board is shown in figure 3.2. Four 8708 ultraviolet erasable read only memory chips are used together with an 8205 1 of 8 decoder for address decoding. Bits 0 to 9 of the address bus are connected to all the memory chips and bits 10, 11 and 12 are fed to the decoder. Hence the memory chips are given the following addresses :

- Device 1 : 000 -- 3FF
- Device 2 : 400 -- 7FF
- Device 3 : 800 -- BFF
- Device 4 : C00 -- FFF

The addresses are specified by the hexadecimal number system. The data bus is driven by the ROM board only when an address within this range is requested and the control line MEMR is low.



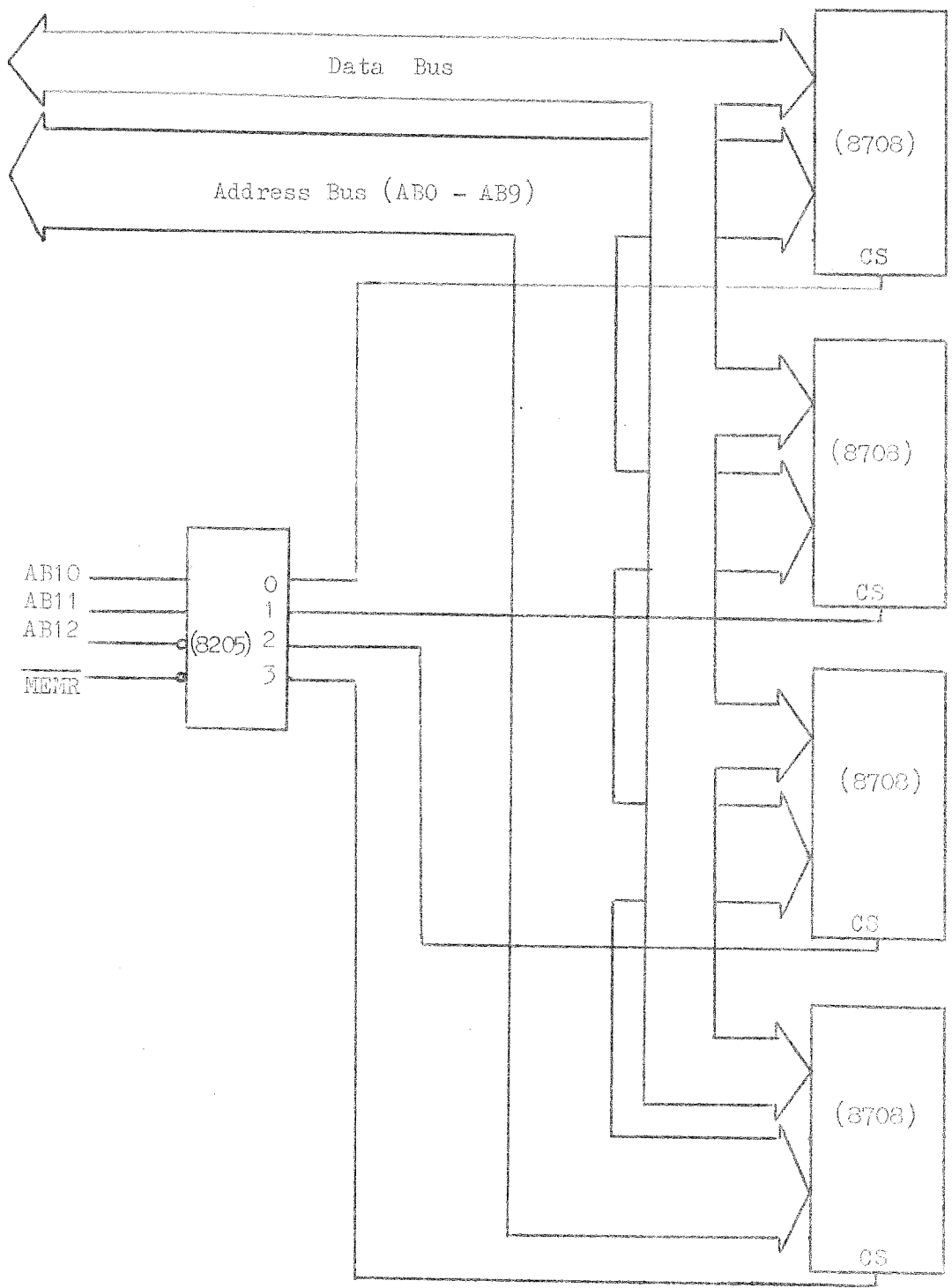


Figure 3.2 : Read Only Memory Board

## 3.3

RANDOM ACCESS MEMORY

The random access memory (RAM) requirement of the analyser is for 2048 words of 8 bit length. This is met by using 16 of the 8111 static MOS RAM (256 X 4) as shown in figure 3.3. Address bits AB0 to AB7 are fed to every RAM chip and each 8 bit word is achieved by using two 4-bit data ports. The 8 pairs of chips are selected by an 8205 1 of 8 binary decoder. This device is controlled by address bits AB8 to AB12. By this means 2K of RAM is available between addresses 1000 to 17FF (hexadecimal).

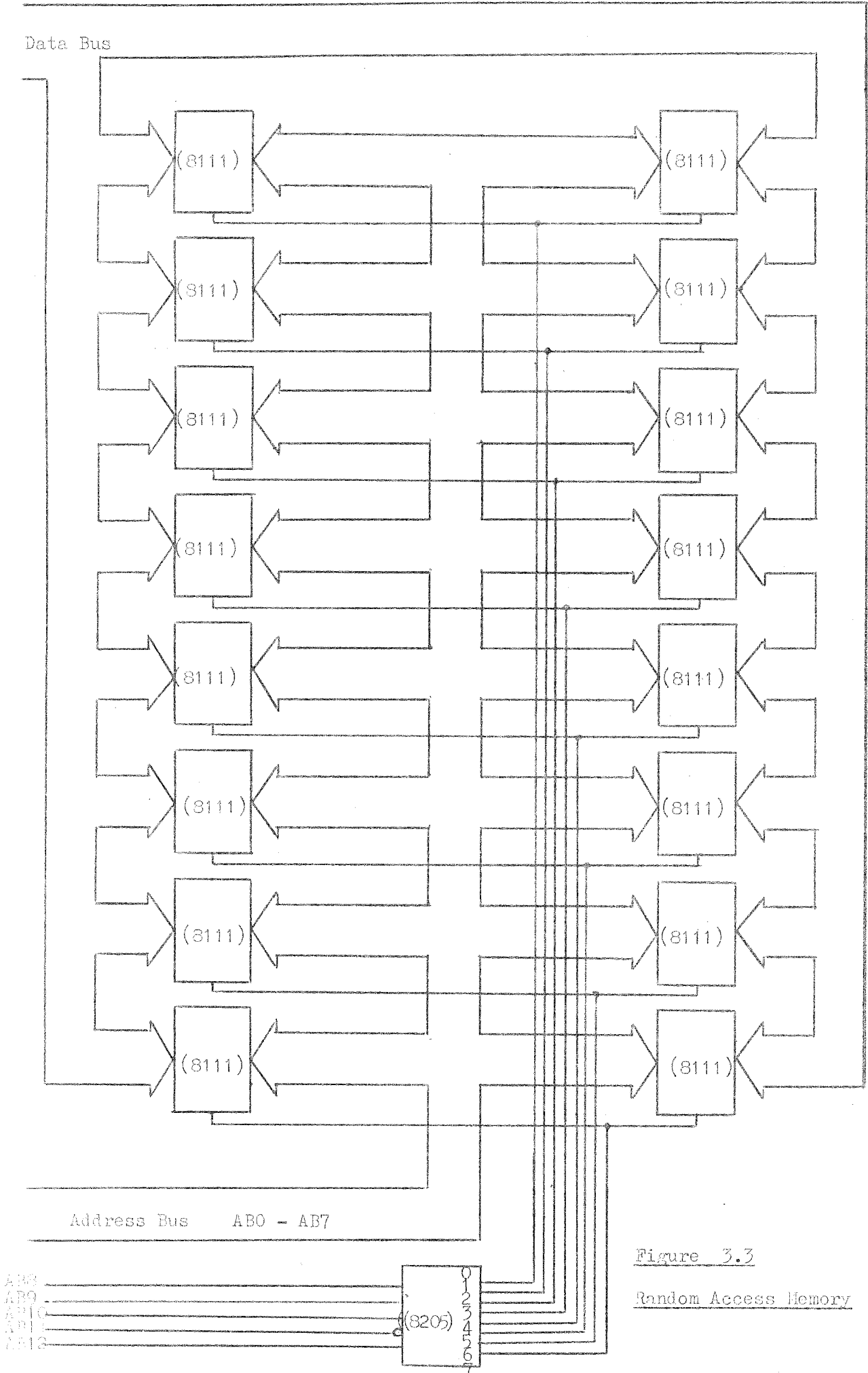


Figure 3.3  
Random Access Memory

3.4

TELETYPE COMMUNICATIONS BOARD

The final spectrum analyser runs without teletype control and at that stage the teletype communications board is redundant. However during the development stage the analyser is controlled by a teletype terminal using a 110 baud serial communication link. An 8251 Programmable Communication Interface integrated circuit and a level translator (built from discrete devices) is used to convert the 8 bit TTL level data bus bits into a serial format and voltage level suitable for operating with a teletype. This board is removed in the final product.

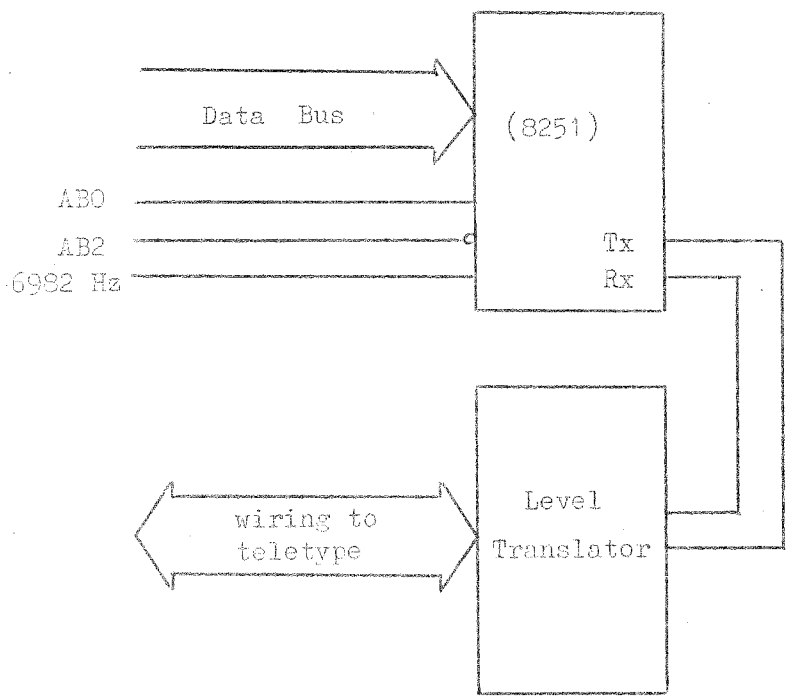


Figure 3.4 : Teletype Communications Board

HARDWARE MULTIPLIER BOARD

Because of the large number of multiplications that are required to be done in the spectrum analyser the overall execution time can be reduced by a factor of 10 to 20 by using a hardware multiplier rather than using a software multiplication routine. The heart of the multiplier is a pair of Am 25LS14 Serial/Parallel Two's Complement Multiplier integrated circuits. The Am 25LS14 is an 8 bit by 1 bit sequential logic element that performs digital multiplication of two numbers represented in two's complement form to produce a two's complement product without correction by using Booth's Algorithm internally.

The 16 bit multiplier is loaded in parallel into two 8 bit shift registers (DM 74166). The 16 bit multiplicand is held at the two 8-bit multiplicand inputs of the AM 25LS14's and the multiplier is fed in serially. The result comes out serially and is fed into three 8-bit serial/parallel registers (Am 25LS22). The multiplier could produce a 32-bit result but only the 24 most significant bits are required. Data flow between the multiplier and the microprocessor data bus is achieved via input/output ports. The timing and control circuitry is achieved by 2 counters (DM 74161) and some logic and the multiplication routine is initiated every time there is an input/output/write operation by the microprocessor. The multiplication time is less than 4 usecs. Faster all-parallel multipliers are available but the microprocessor input/output operations are too slow to take advantage of them.

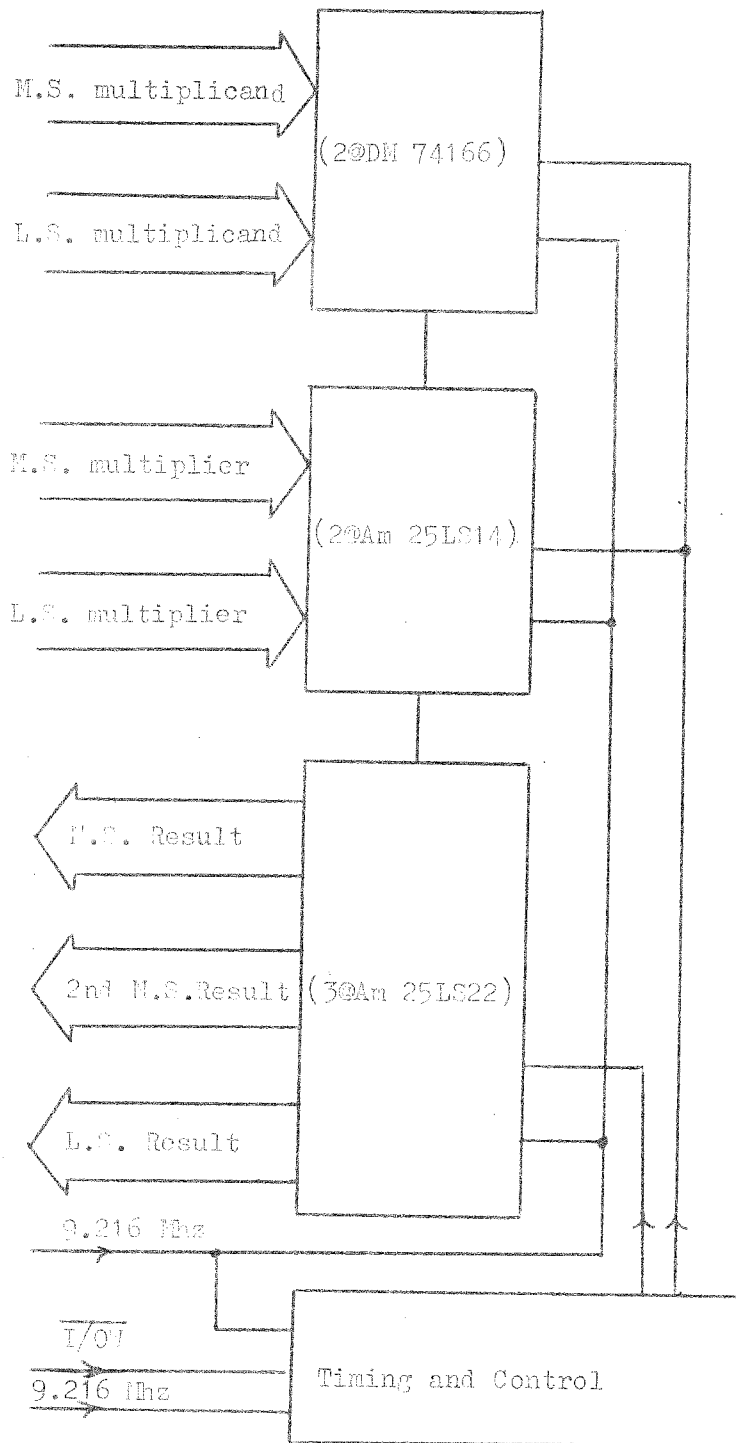


Figure 3.5 : Hardware Multiplier Board

3.6

SAMPLE RATE GENERATOR

The purpose of the Sample Rate Generator is to produce several clock frequencies which are used by other sections of the analyser for timing purposes. The Generator as shown in figure 3.6 consists of a crystal oscillator which drives a chain of counters which provide frequency division. The 1.6384 Mhz crystal frequency is divided by 2 to give an 819.2 Khz frequency which is used as the clock for the successive approximation register (SAR) of the A/D converter.

The frequencies : 256 Hz to 12.8 Khz are selected to control the sampling rate of the A/D converter. The 256 Hz signal is also used to control the writing rate for the output recorder.



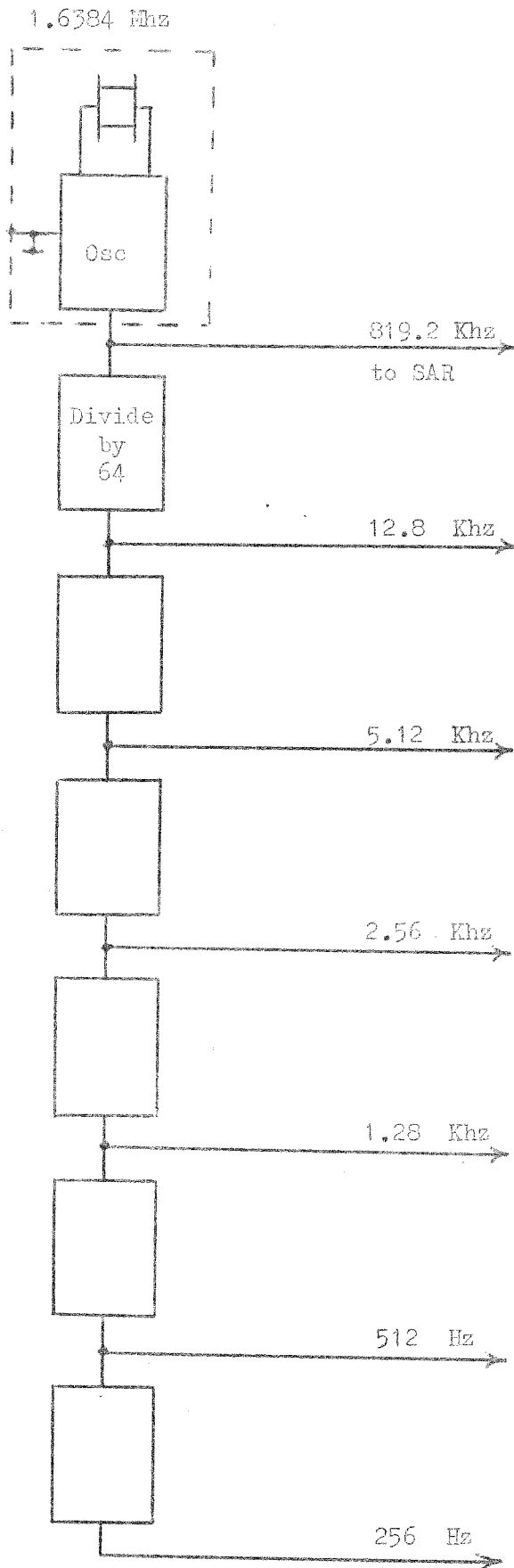


Figure 3.6 : Sample Rate Generator

## 3.7

INPUT AMPLIFIER (figure 3.7)

The input amplifier is required to amplify very low level signals on the high gain range and to keep noise levels down several precautions have been taken :

- (a) The operational amplifier used is an OP-01 low noise device.
- (b) Ground leads for digital and analogue devices have been separated to eliminate 'cross-talk'.
- (c) The complete amplifier assembly (including the selector switch and feedback resistors) are enclosed in a shield which is connected to the body of the BNC connector to which the input leads are connected.
- (d) The BNC connector is connected to ground only through the input return lead to prevent ground circulating currents.

The output of the amplifier is fed via the first bank of frequency selector switches to the input low pass filters.

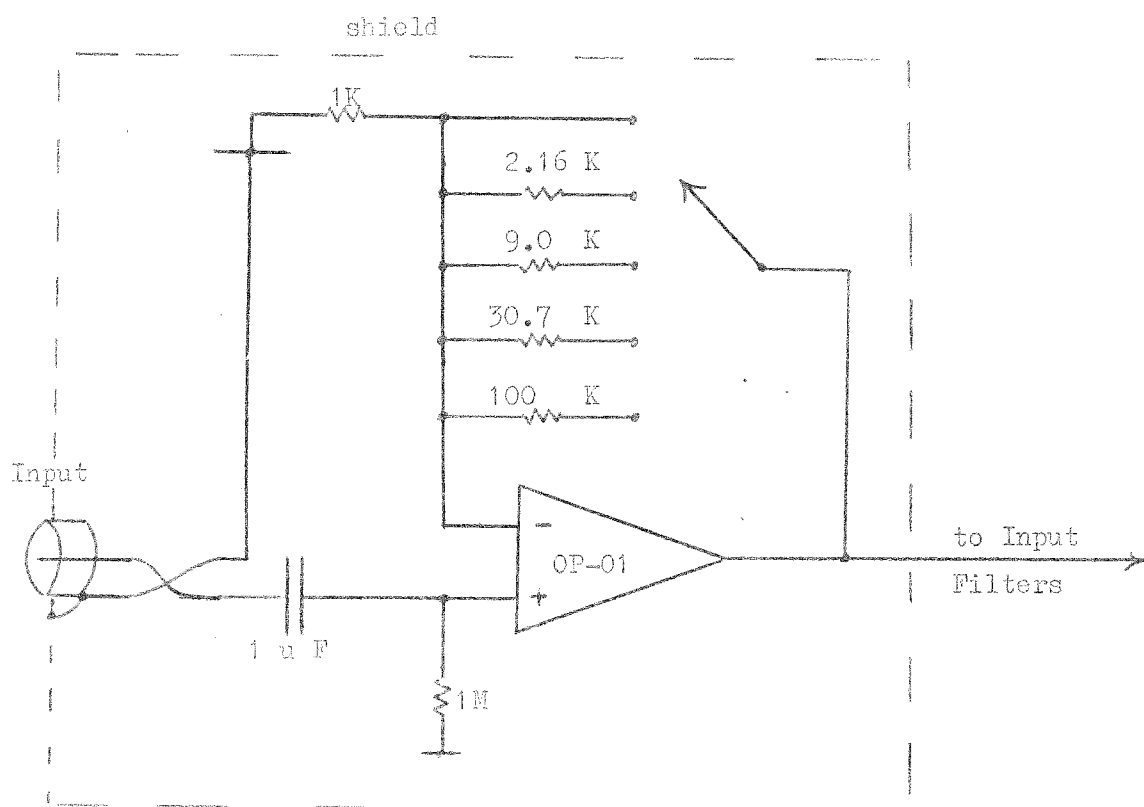


Figure 3.7 : Input Amplifier

The purpose of the input filters is to attenuate any frequencies above half the sampling rate. The filters are low pass 5th Order Elliptic filters. The transfer function is of the form :

$$T(S) = \frac{a_4 S^4 + a_2 S^2 + 1}{b_5 S^5 + b_4 S^4 + b_3 S^3 + b_2 S^2 + b_1 S + 1} \quad \text{where } S = j\omega$$

The 'Handbook of Filter Synthesis' by Anatol Zverev was used to select the poles and zeroes of this transfer function and these are listed below for a 1 radian/second cut-off frequency :

Poles :

$$S = -0.5573$$

$$S = -0.10672 \pm j1.0457$$

$$S = -0.37187 \pm j0.7375$$

Zeroes :

$$S = \pm j2.4377$$

$$S = \pm j1.617$$

The stopband begins at 1.56c/sec and there is greater than 48 db attenuation in the stop band. The overall response is shown in figure 3.8(a).

To obtain the required transfer function  $T(S)$  is expressed as the product of 2 biquadratic functions and a single pole :

$$\text{ie. } T(S) = \frac{.55734}{S + .55734} \times \frac{S^2 + 5.924}{S^2 + 0.21344S + 1.10488} \times \frac{S^2 + 2.6147}{S^2 + .74374S + 0.68219}$$

Each biquadratic transfer function is simulated by the B1-QUAD active filter arrangement as shown in figure 3.8(b) and the single pole is realized by an R-C filter.

The operational amplifier used is the LM324 Quad OP-AMP. Hence 2 of these are required for each filter.

## 3.8(cont.)

There are 6 filters wired on 2 boards which are selected by the frequency selector switches. The cut-off frequencies and corresponding sampling rates are listed below :

Cut-off Frequency (Hz) $f_c$	Sampling Rate (Hz)
100	256
200	512
500	1280
1000	2560
2000	5120
5000	12800

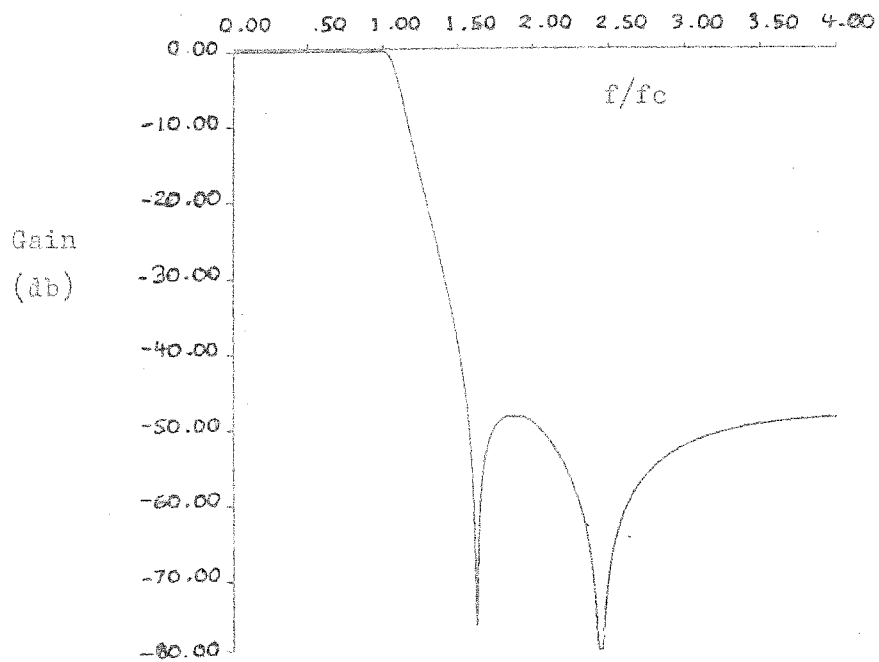


Figure 3.8(a) : Transfer Function of Input Filters

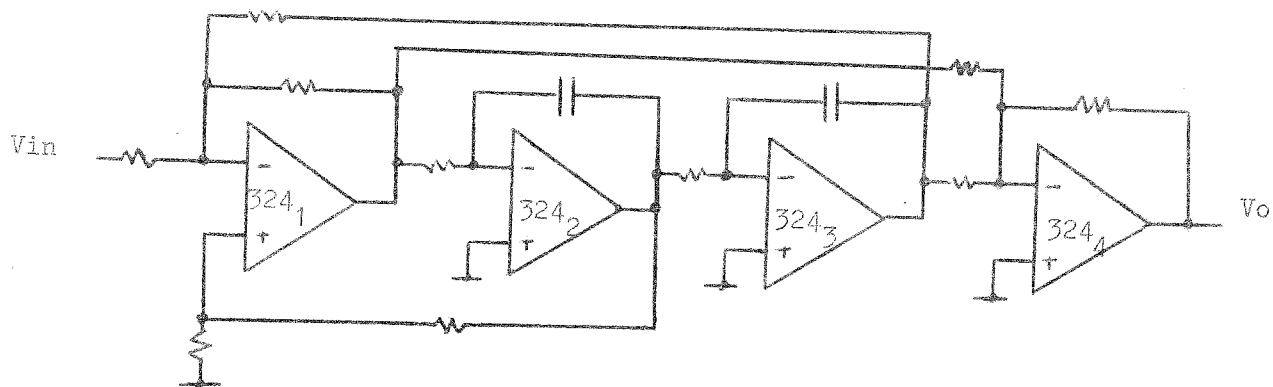


Figure 3.8(b) : Bi-Quad Active Filter

## 3.9

ANALOGUE TO DIGITAL CONVERSION

The analogue signal received from the input filters is sampled and digitized by an analogue to digital converter. The A/D converter is a conventional successive approximation type of converter consisting of a D/A converter, a comparator and a successive approximation register (SAR) as shown in figure 3.9. The result is a 10-bit two's complement binary number at the completion of each conversion. The SAR is clocked by the 819.2 Khz clock from the sample rate generator board and each conversion is initiated by the sampling frequency. The sample and hold device (S/H) samples the signal until conversion commences and holds the signal at a constant voltage until after the conversion is complete. The 10-bit result is read by an input-output port (8255).

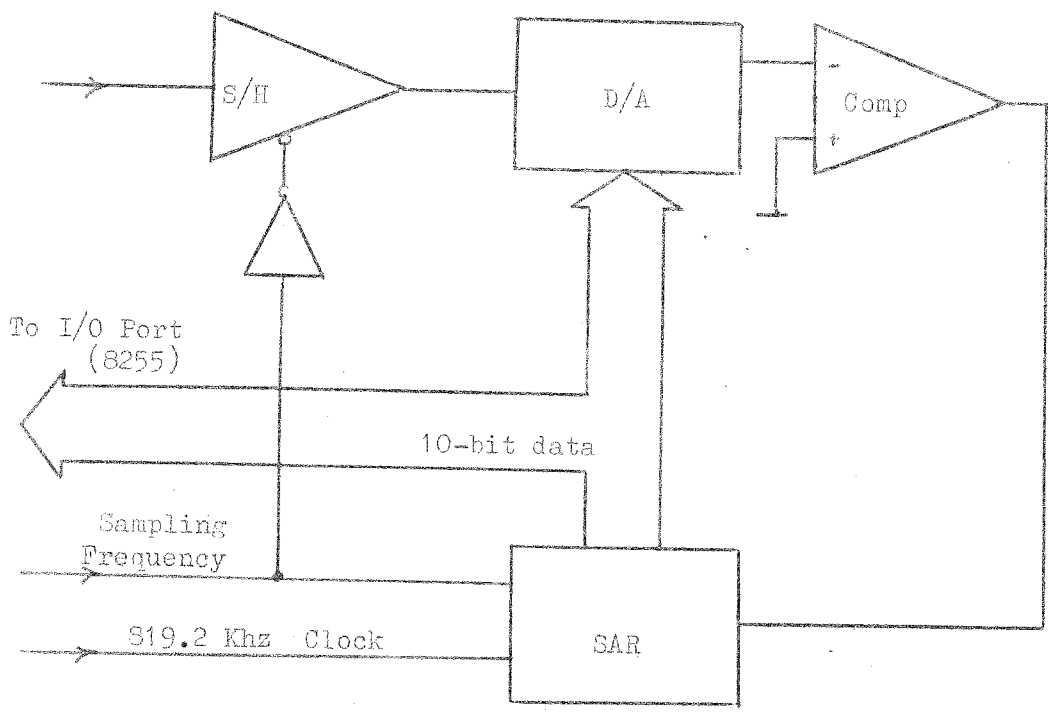


Figure 3.9 : Analogue to Digital Conversion



DIGITAL TO ANALOGUE CONVERSION

The spectrum analyser displays the spectra on a CRO or a paper recorder. To do this the numerical values calculated by the microprocessor must be converted into analogue voltages which is achieved by digital to analogue conversion as shown in figure 3.10.

The 8-bit digital number is fed from the microprocessor data bus to a D/A converter (DAC100) via an input/output port (8255). The DAC100 produces a current output, which is converted to voltage by an operational amplifier. This voltage is fed to a resistor attenuator (used for scaling) and 4 CMOS switches direct the voltage to either the CRO output or the recorder output. The state of the switches is controlled by the microprocessor via the input/output port.

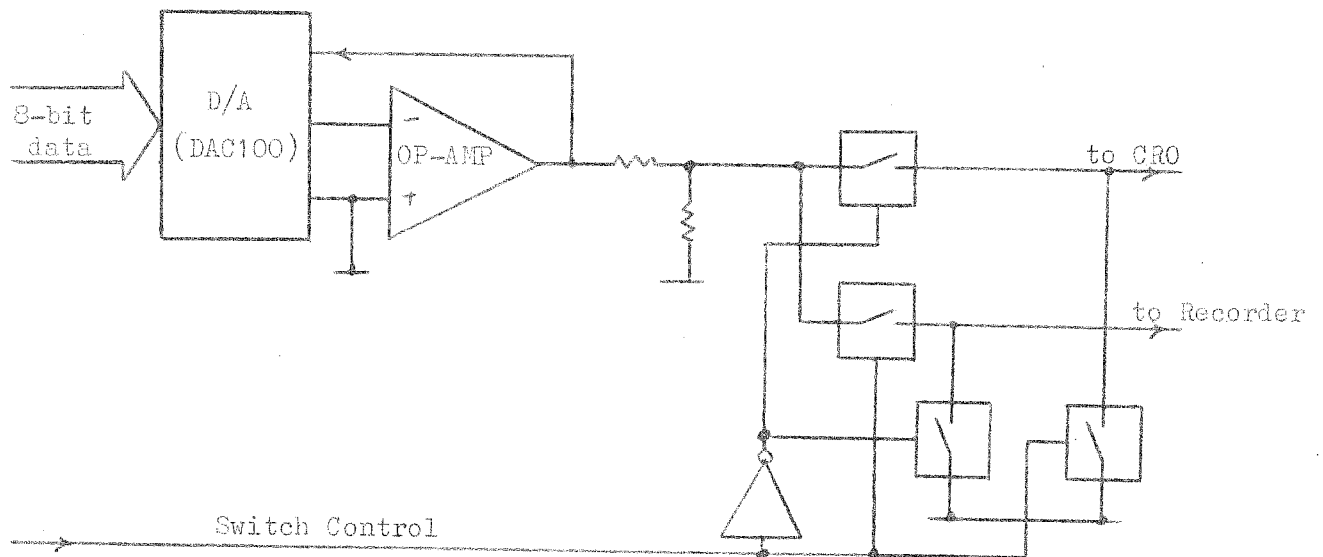


Figure 3.10 : Digital to Analogue Conversion

FREQUENCY DISPLAY AND CURSOR CONTROL

To facilitate reading of the spectrum on the CRO two signals are sent to the CRO ie. the spectrum and the display trigger. The CRO is triggered from the negative edge of the display trigger and the positive edge (also displayed on the CRO) corresponds to the indicated frequency on the frequency display. The frequency display (figure 3.11) consists of 4 seven-segment displays which are driven by 4 BCD-7 segment decoders which receive their information from the microprocessor via an input/output port (8255). The position of the frequency selector switch and the state of the cursor control switch are read by the same input/output port.

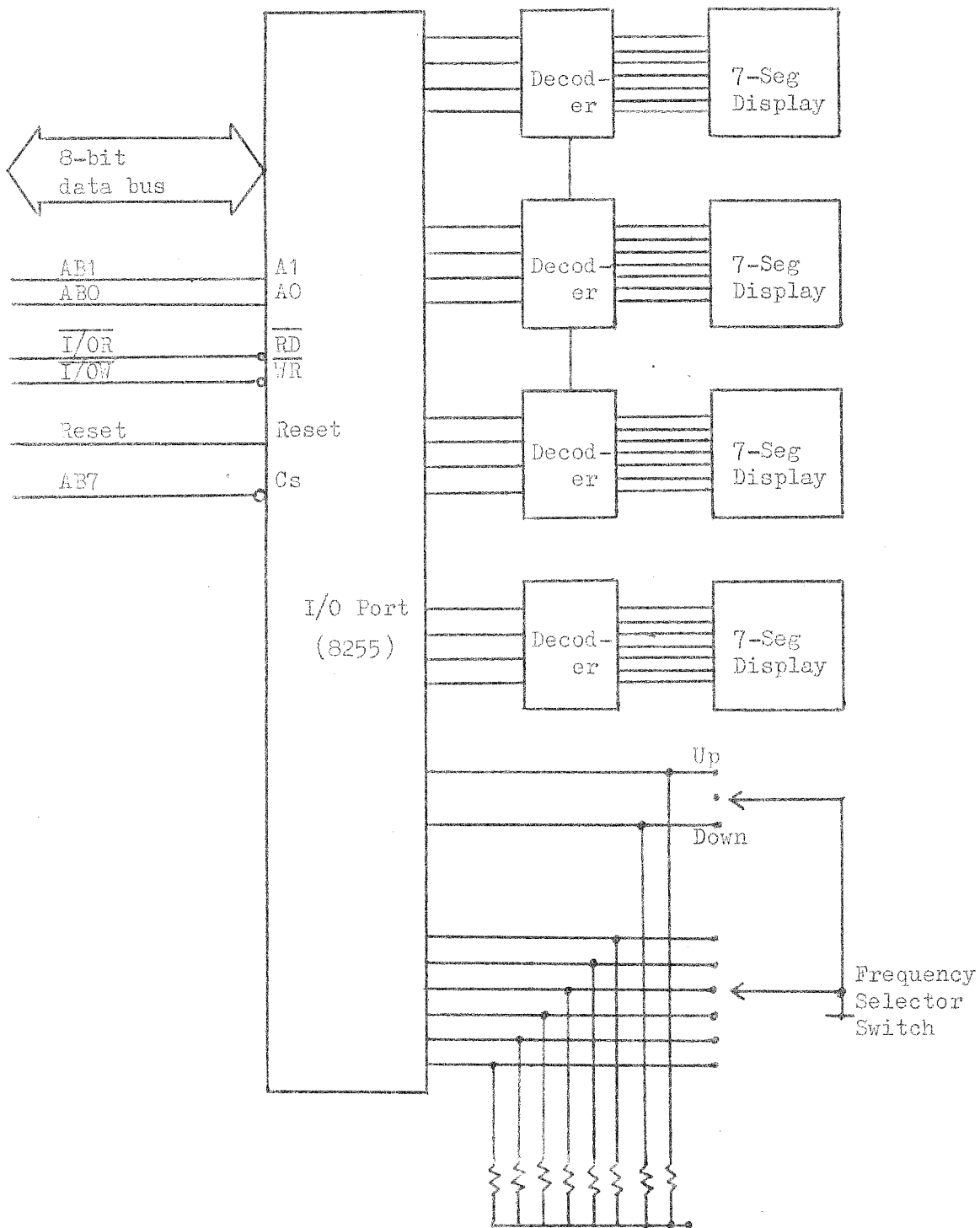


Figure 3.11 : Frequency Display and Cursor Control

POWER SUPPLY

The power supplies are a conventional series regulated design where the regulators are integrated 3 terminal devices. The  $\pm 12$  and 5V supplies are for digital circuitry and the  $\pm 15$  V supplies are for analogue circuitry. The ground lines for the digital and analogue circuitry are kept separate to prevent 'cross talk'.

REFERENCE

- [1] "Digital Signal Processing Handbook"  
by John R Mick, Advanced Micro Devices, 1976.
- [2] "8080 Microcomputer Peripherals User's Manual"  
INTEL Corporation 1976.
- [3] "INTEL 8080 Microcomputer Systems User's Manual"  
INTEL Corporation, September 1975.

SOFTWAREDEBUGGING TECHNIQUE

In practice the spectrum analyser is to be controlled by a reset button which initiates a calculation cycle. The first instruction executed is stored at location zero and is the beginning of the main program. This set-up however is unsuitable as a means of development as it provides no means for interrogating the inevitable mistakes.

To provide an interrogation facility a 'monitor' program, stored on an EPROM, is placed in the first K (1024 words) of memory which enables a teletype terminal to control the microprocessor. This facility allows hexadecimal memory data to be stored, displayed or shifted and sections of program can be tested individually.

The EPROM (Erasable Programmable Read Only Memory) containing the main program is placed in the third K of memory and has a few 'RST 1' instructions which cause program execution to jump back to monitor program which restores control to the teletype. Some of the subroutines are also stored in the same EPROM as the main program. The remainder of the subroutines and the sine look-up table are stored in another EPROM occupying the second K of memory. The 4th K of memory was also loaded with the same program as the 3rd except that only one 'RST 1' instruction was used and inserted at the end of the main program so that the whole program can be tested at once yet retaining teletype control.

PROGRAMMING OF EPROMS

The program was originally written in INTELL 8080 assembly language and punched onto cards. The assembly language was converted into machine code (hexadecimal) using a library routine (INTELLEN) available on the university CDC Cyber 173 computer. The machine code was punched onto paper tape which enabled the code to be loaded onto a PROMPT80 which is an instrument manufactured by INTELL to be used as a teaching aid but also has a facility for programming the 8708 EPROM. There are many other ways to program EPROMS but the way chosen was the most convenient for the given situation.

## 4.3

ALLOCATION OF RANDOM ACCESS MEMORY (RAM)

The spectrum analyser uses 2 K words of RAM (ie. 2048) which is located in the hexadecimal address range 1000 to 17FF. The way in which each section of RAM is used is illustrated in figure 4.1. The Accumulated Power Spectral Density is stored in locations 1000 to 12FF and is represented by PSD (L) where  $L = 0 \dots 255$ . The area of memory allocated to FFT calculations is 1400 to 17FF.

$X(L)$  = real part of array.

$Y(L)$  = imaginary part of array.

Each PSD (L) entry consists of 3 bytes (ie. 24 bits).

Each X (L) and Y (L) entry consists of 2 bytes (ie. 16 bits).



(1000)	Most Significant byte of PSD (L)
(10FF)	- - - - -
(1100)	2nd M.S. byte of PSD (L)
(11FF)	- - - - -
(1200)	3rd M.S. byte of PSD (L)
(12FF)	
(1300)	Output Display Array
(137F)	
(1380)	General Purpose
(13FF)	
(1400)	
	M.S. byte of X(L)
(14FF)	- - - - -
(1500)	
	L.S. byte of X(L)
(15FF)	
(16FF)	
	M.S. byte of Y(L)
(16FF)	- - - - -
(1700)	
	L.S. byte of Y(L)
(17FF)	

Figure 4.1: Allocation of RAM

NUMBER REPRESENTATION

Two types of number representation are used :

(a) Integer Representation

This representation is used most frequently for counters controlling the flow of programs.

eg.           0000 1000 = 8

(b) Twos Complement Representations

This representation is used for most of the values and arrays which are the object of the computations (ie. X (L), Y (L), and PSD (L)).

The range of possible values are  $(-\frac{1}{2}, +\frac{1}{2})$ .

The most significant bit of a word indicates sign.

eg.           1 000       =  $-\frac{1}{2}$

              0 100       =  $+\frac{1}{4}$

THE MAIN PROGRAM

The flow chart for the main program is shown in figures 4.2 (a) and 4.2 (b) and the actual coding for the main program as well as the subroutines and sine look-up table is listed in Appendix 2.

The function of the main program is to:

- (1) initialize input-output ports and the microprocessor.
- (2) read the state of front-panel switches.
- (3) set up initial conditions.
- (4) control the calling of subroutines.

Some of the switch conditions are read once only per calculation routine ie. Window Selection, no. of accumulations and bandwidth selection are not able to be changed until the next calculation since otherwise the result would be meaningless. However switch positions which control the mode of displaying the result such as CRO or recorder output, or logarithmic or linear output, are regularly monitored for a change of state.

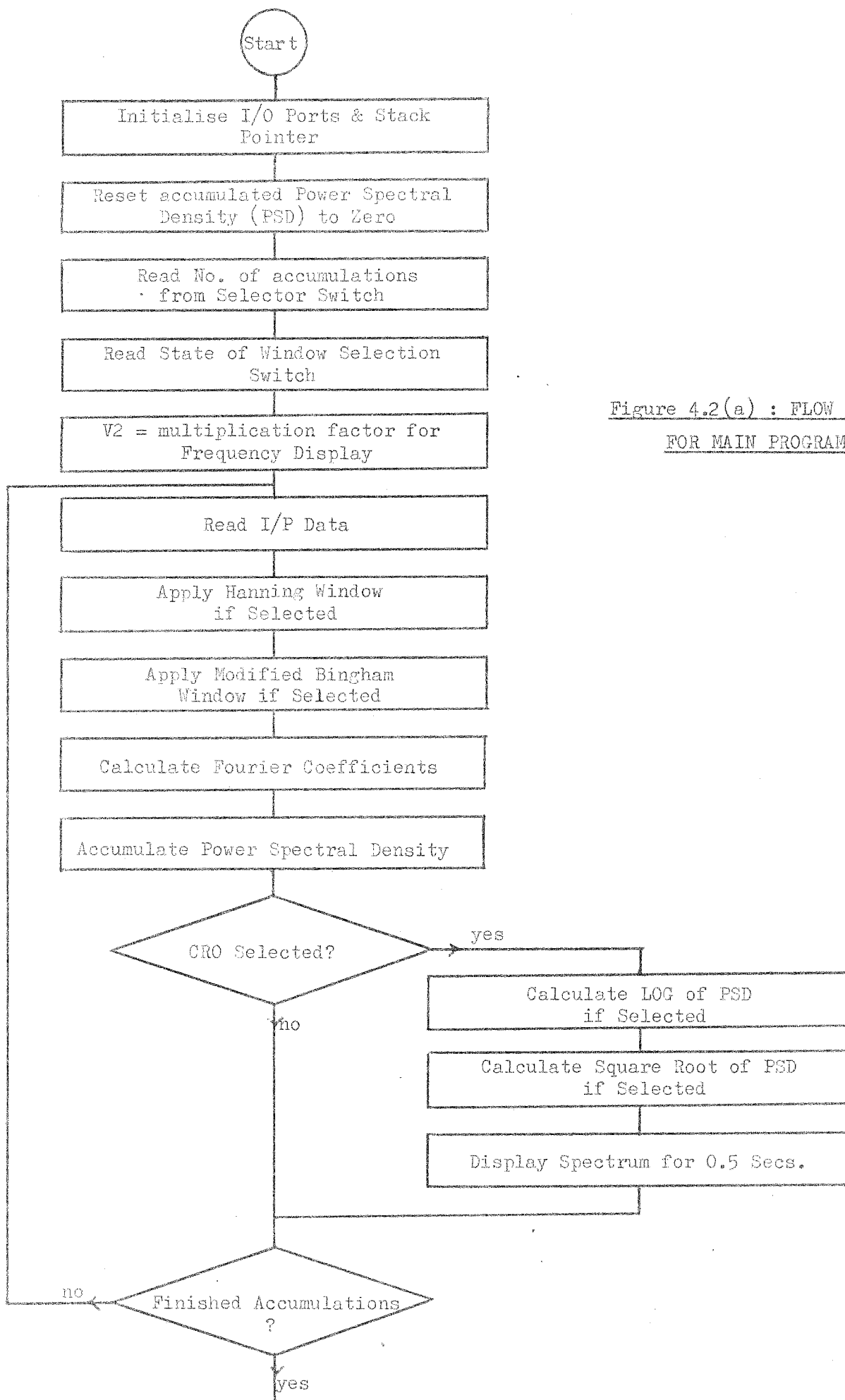


Figure 4.2(a) : FLOW CHART  
FOR MAIN PROGRAM

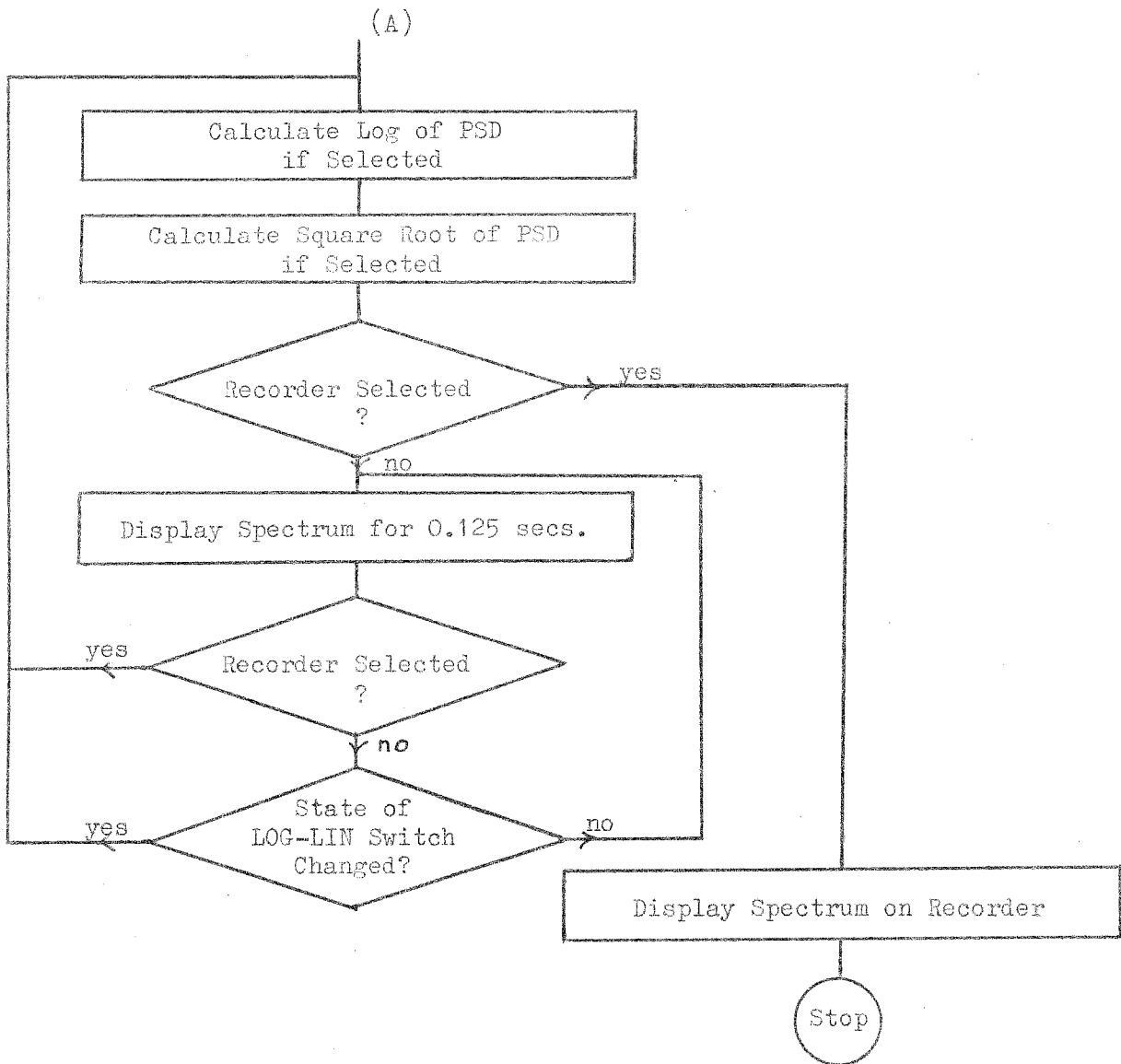


Figure 4.2(b) : FLOW CHART FOR MAIN PROGRAM

INPUT ROUTINE

The Input Routine is a subroutine called from the main program and performs the following functions (refer to figure 4.3):

(a) Sample Input Signal

A clock signal selected from the sample rate controller board is monitored by an I/O port and the analogue to digital (A/D) converter is read 15 to 20 usecs after a low to high (LO-HI) transition. This clock signal initiates an A/D conversion and the time delay allows enough time to complete the conversion. The program reads 256 10-bit samples (X (L)). The most significant 8 bits are stored in locations 1400 to 14FF and the remaining 2 bits are stored in locations 1500 to 15FF. (The address locations are expressed in the hexadecimal number base as will be assumed for the rest of the text).

(b) Store Zeroes in Imaginary Area of Memory

The fast fourier transform operates on complex data and so operates on arrays of memory which consist of real and imaginary parts. The transform is to operate on the input signal which is a real array and so the area of memory allocated to imaginary data is set to zero prior to the FFT being executed. This area of memory comprises locations 1600 to 17FF.

(c) Check Input Data for Overload

If the input signal is too large for the A/D converter to handle then the numbers read will be limited to positive or negative maximum. Hence during this section of the program the most significant byte of each sample is tested for equality with the hexadecimal numbers 7F (positive maximum) or 80 (negative maximum).

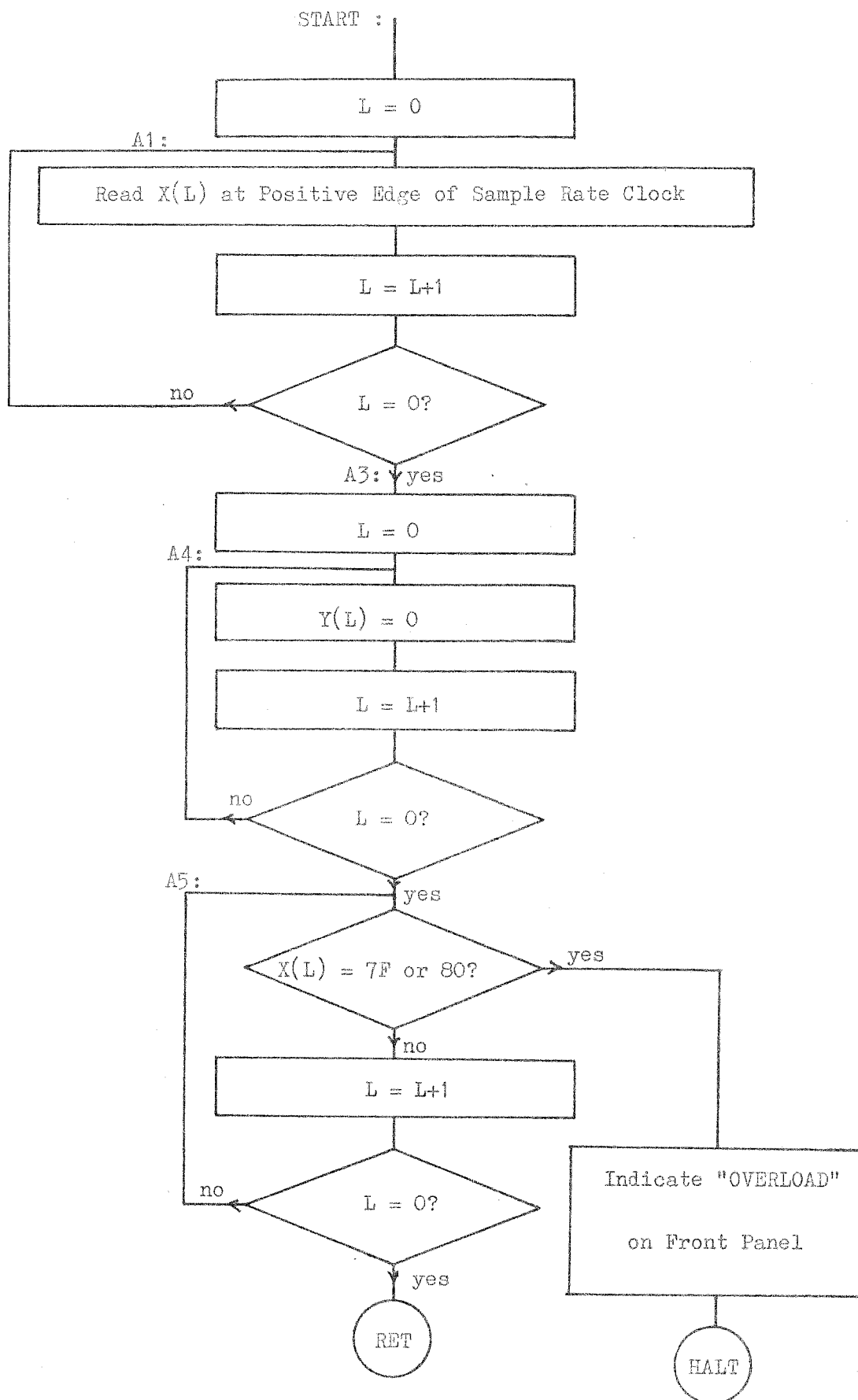


Figure 4.3 : Input Routine

If this condition is met the appropriate LED on the front panel is energised and the program is stopped. If overload has not occurred then execution is returned to the main program.



PREMULTIPLICATION OF INPUT DATA BY A WINDOW FUNCTION

As mentioned in section 2.3, the spectrum can be improved by premultiplying the input signal by certain functions over the sample range. Three window functions can be selected by a 3 position switch on the front panel which is interrogated by the main program. These 3 functions are :

- (a) Rectangular Window (ie. No Window)
- (b) Modified Bingham Window
- (c) Hanning Window

The Rectangular Window simply describes the case of not applying any window ie. the input data is unmodified. The remaining 2 window functions have been described in section 2.3 and are achieved in practice by the microprocessor executing the flow charts shown in figures 4.4 and 4.5 respectively. These flow charts are given in terms of machine registers A, B, C, D, E, H and L. In interpreting the flow charts these registers are treated as integers of moduli 256. The sine look-up table is represented by  $S(I) = \sin(2\pi I/256)$  for  $I = 0 \dots 255$ . This convention is adhered to for all remaining flow-charts.

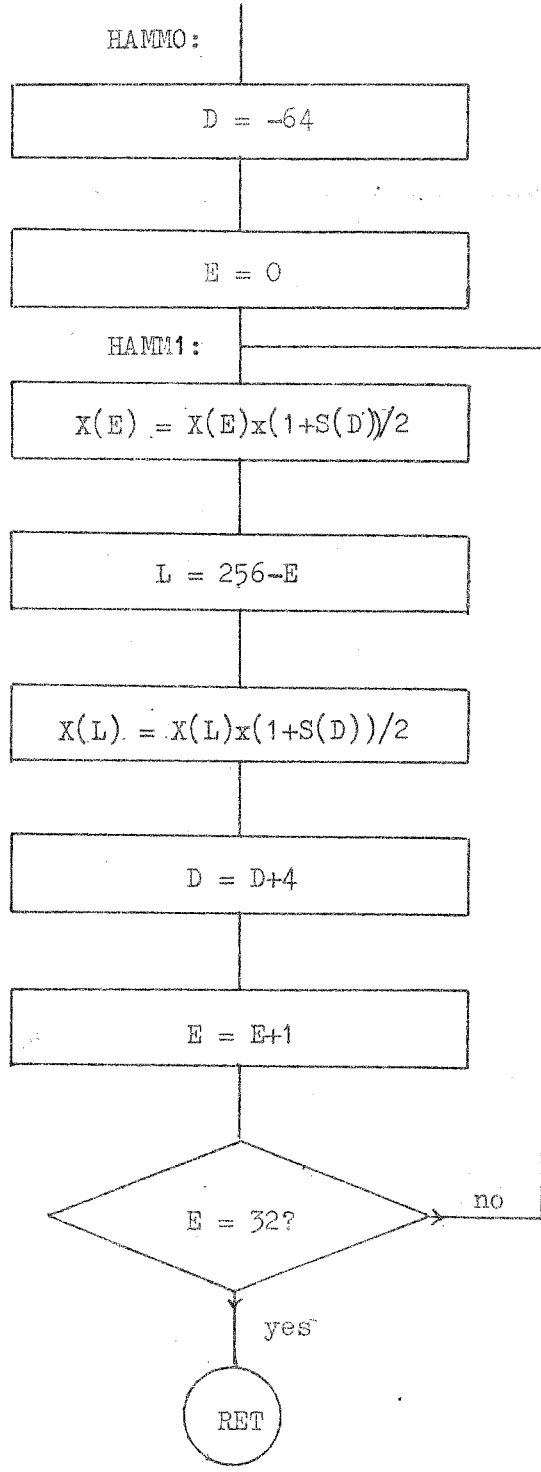


Figure 4.4 : Modified Bingham Window

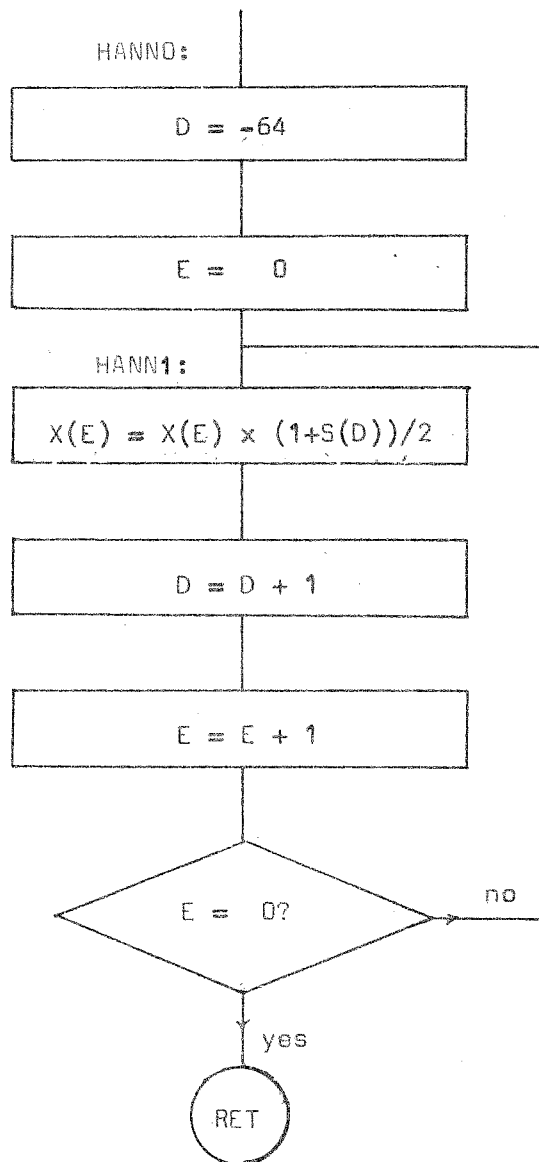


Figure 4.5 : Hanning Window

## 4.8

CALCULATION OF FOURIER COEFFICIENTS

The DFT of the input data is calculated using the Fast Fourier Transform (FFT) technique. This is achieved by 2 subroutines which are :

- (a) Reshuffle data in bit reversed order
- (c) FFT using decimation in time

## 4.8(a)

RESHUFFLE DATA IN BIT REVERSED ORDER

This subroutine is simply preparatory for the FFT decimation in time routine. It involves rearranging the data in bit reversed order.

Eg. consider  $X(3)$

In binary representation (8 bits) 3 is represented by :

$$0000\ 0011 = 3$$

If this number is bit reversed the result is :

$$1100\ 0000 = 192$$

Therefore the 2 data points  $X(3)$  and  $X(192)$  are interchanged. This process is carried out for all 256 data points. The flow chart for this process is shown in figure 4.6.

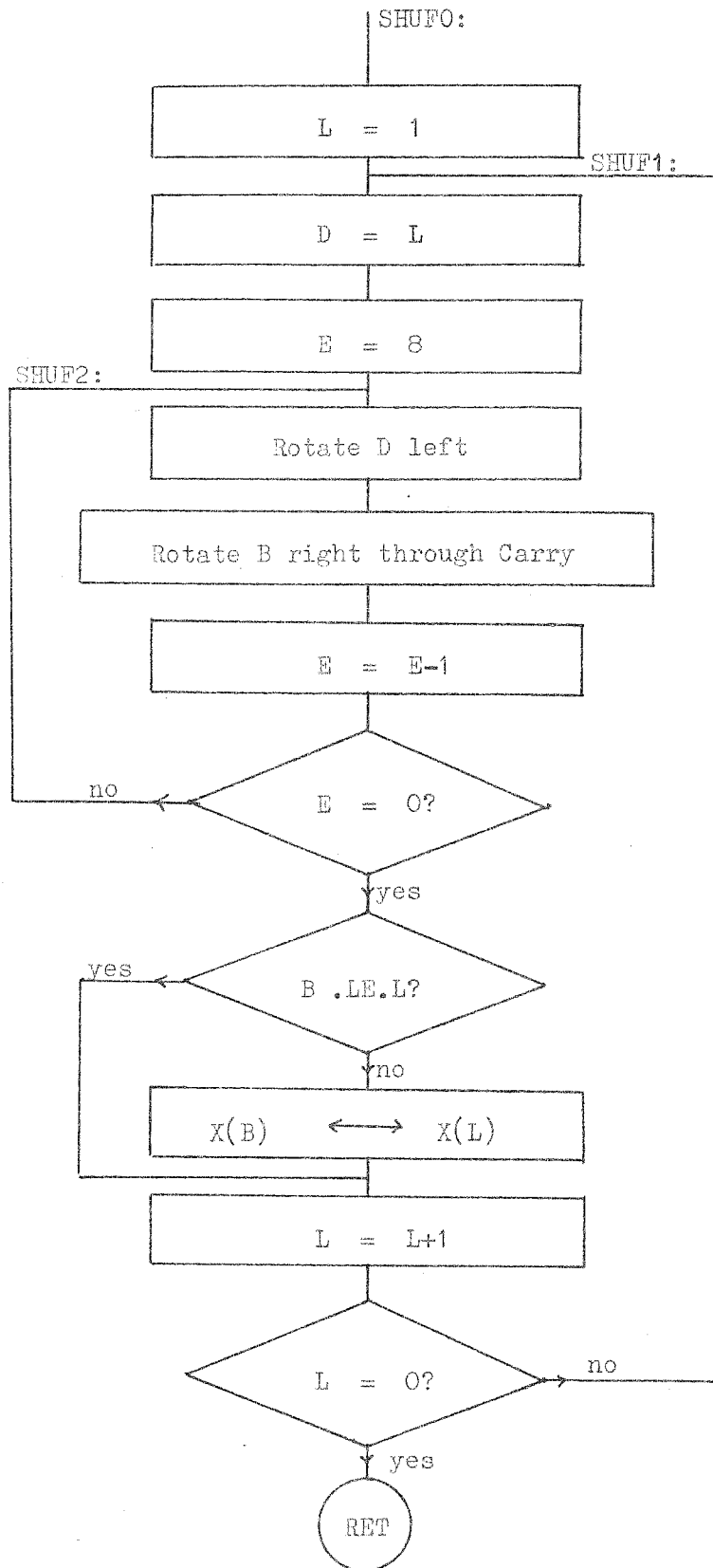


Figure 4.6: Reshuffle Data

4.8(b)

FFT DECIMATION IN TIME

The flow chart for this routine is shown in figure 4.7. The notation in the flow chart is according to the following convention

$X(I)$  = real part of array entry I.

$Y(I)$  = imaginary part of array entry I.

It was noted in a previous section that :

$S(I) = \sin(2\pi I/256)$        $I = 0 \dots 255$

The cosine  $CO(I) = \cos(2\pi I/256)$  can be calculated by :

$CO(I) = S(I + 64)$  from the sine look-up table.

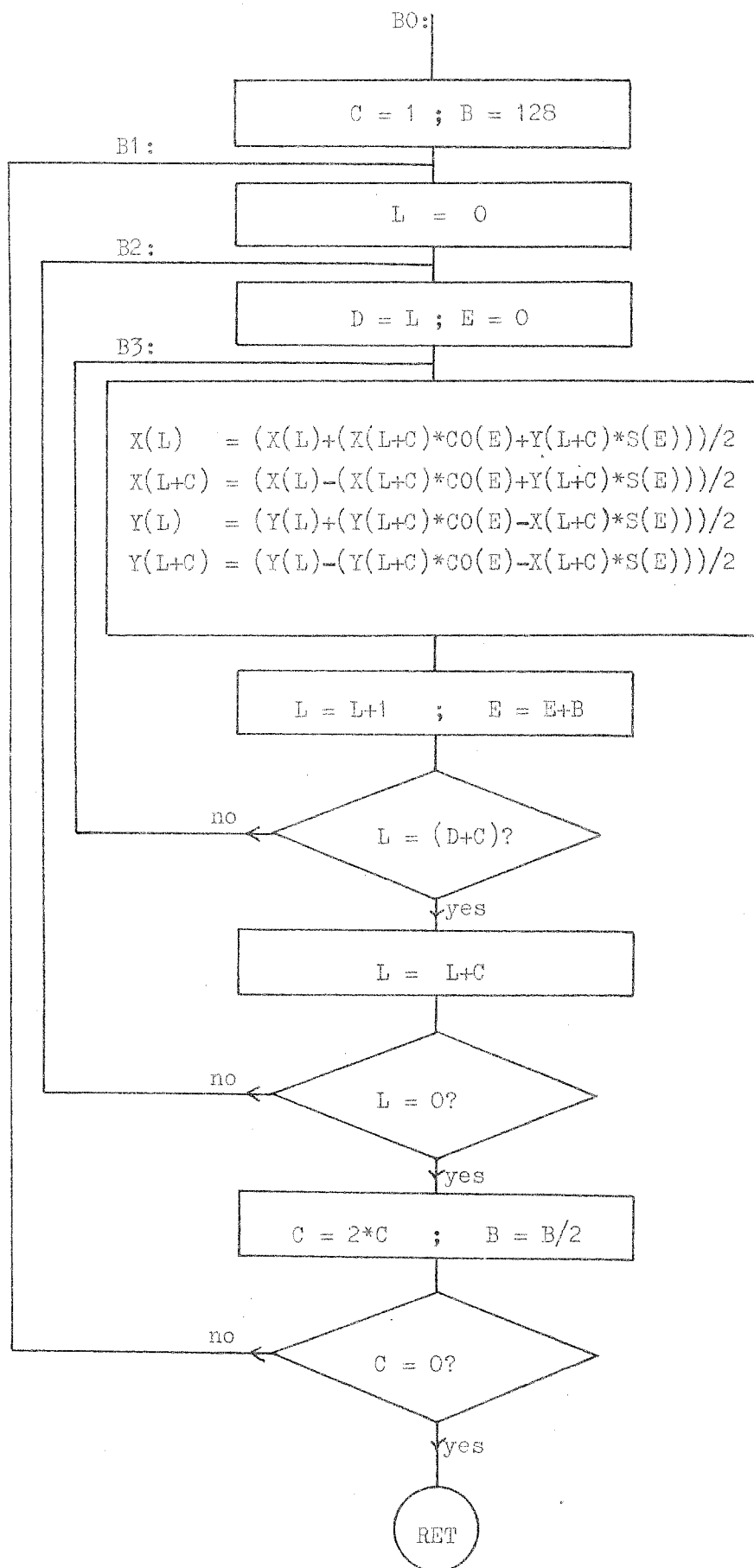


Figure 4.7 : FFT Decimation in Time

AMPLITUDE CORRECTION DUE TO WINDOWING AND ACCUMULATION OF  
POWER SPECTRAL DENSITY

Once the DFT has been calculated, the result is an array which consists of real and imaginary parts. At this stage the DFT components are multiplied throughout by scalar correction factors to compensate for the amplitude error caused by windowing. This correction factor is inversely proportional to the area under the particular window function. The power spectral density array (PSD (I)) is then accumulated by the sum of the squares of the real and imaginary parts of the DFT (X (I) and Y (I) respectively) divided by the total number of accumulations (NA). (NA was read by the main program from the front panel). This process is repeated for each frequency component  $I = 0 \dots 255$ . The flow chart is shown in figure 4.8.



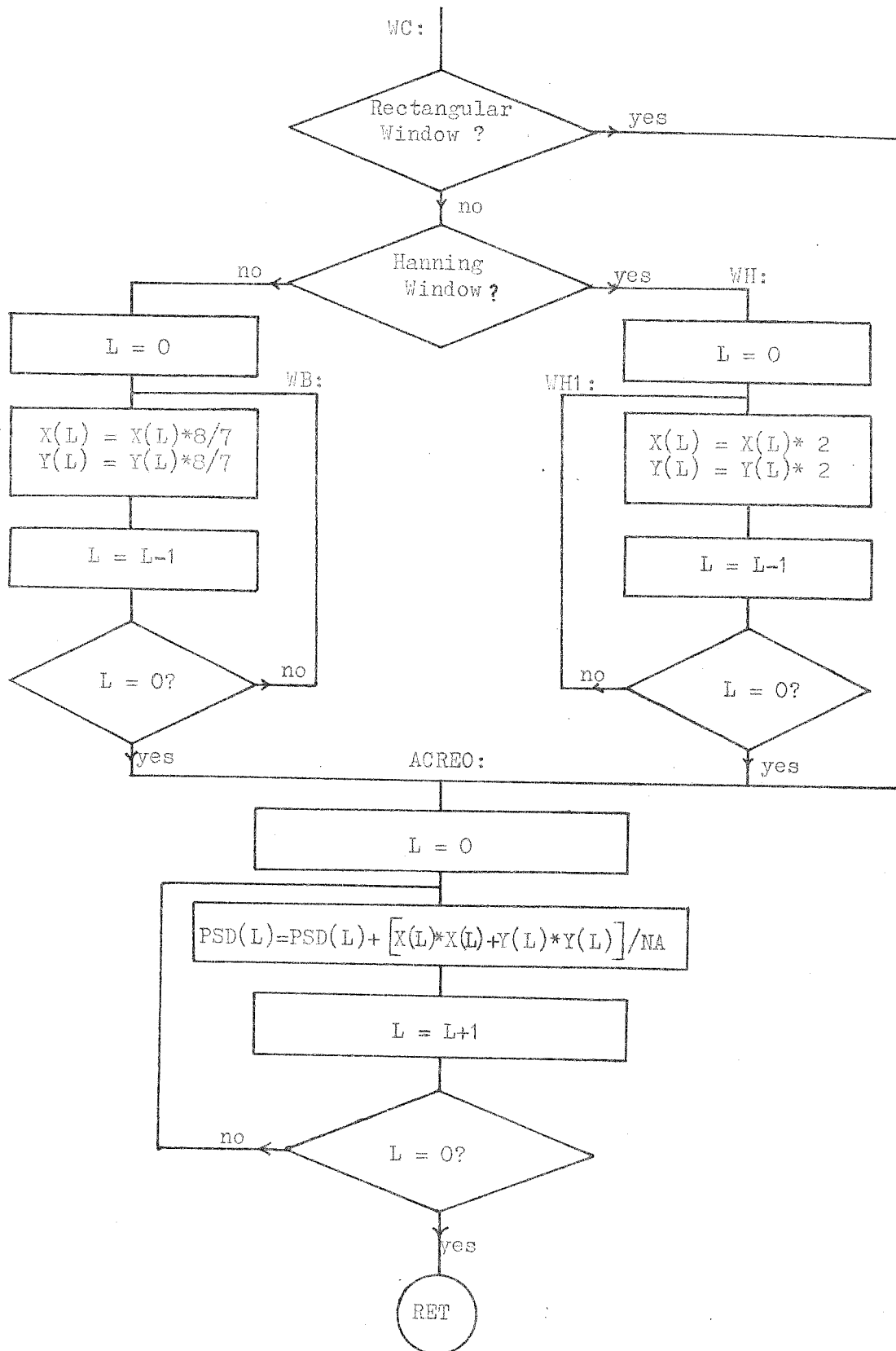


Figure 4.8 : Amplitude Correction due to Windowing and Accumulation of Power Spectral Density

4.10 CALCULATION OF LOGARITHM OF POWER SPECTRAL DENSITY

To display the power spectral density of a signal over a wide dynamic range of amplitude it is necessary to display the spectra on a logarithmic scale. The flow chart for the subroutine 'LOG' is shown in figure 4.9. This program calculates the base 2 logarithm of a 3 byte no. to produce a 3 byte integer result.

ie.  $DIS(L) = 220 + 10 \log_2 (PSD(L))$  where  $L = 0 \dots 127$

An attenuator is set to a specific value in the D/A converter for displaying results so that the final result can be read in decibel units.

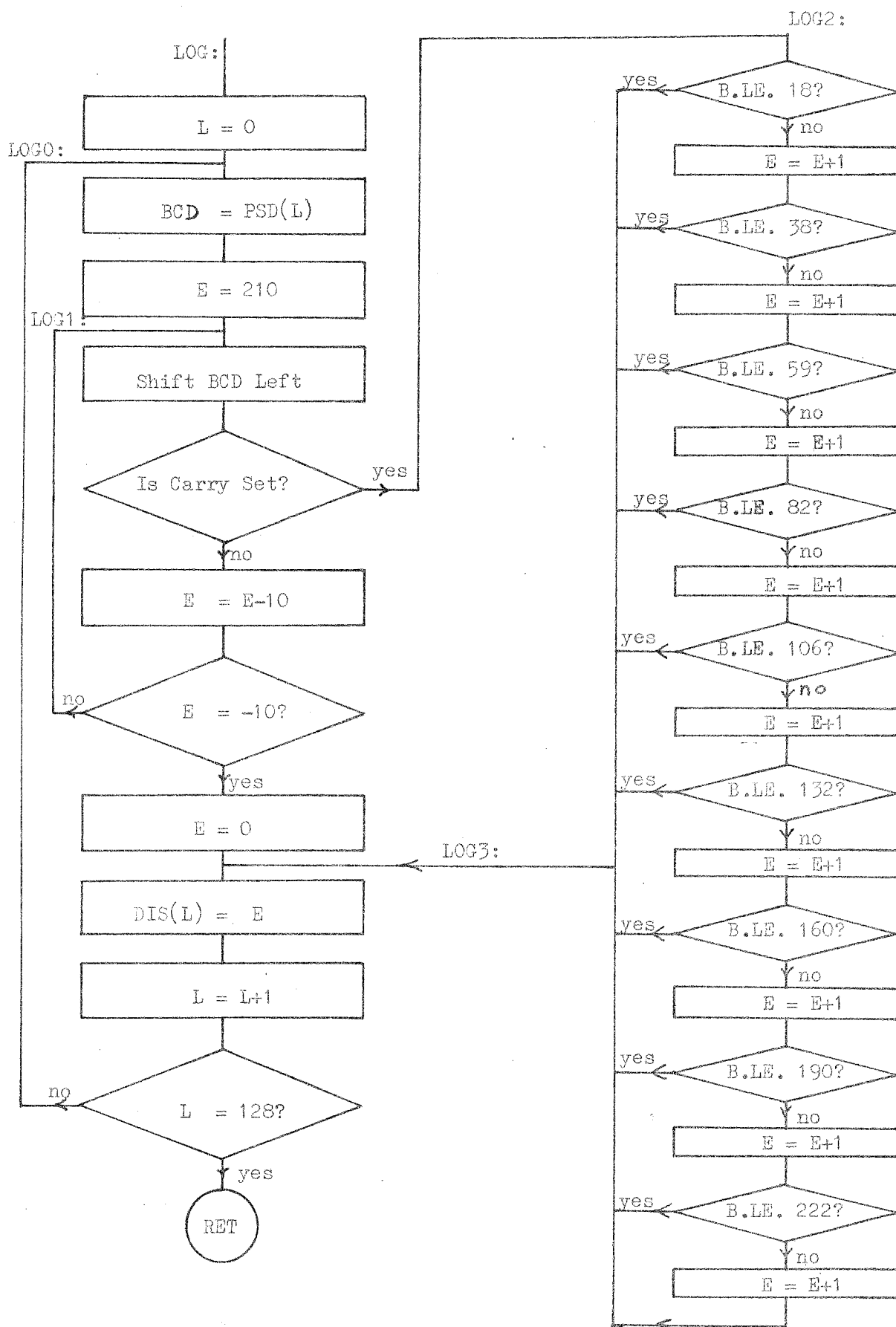


Figure 4.9 : Calculation of Logarithm of Power Spectral Density

4.11 CALCULATION OF SQUARE ROOT OF THE POWER SPECTRAL DENSITY

Quite often it is desirable to display the magnitude of frequency components on a linear scale. This is simply the Pythagorean sum of the result of the FFT operation ie.:

$$\begin{aligned} &= \text{SQRT} (X (L)* X (L) + Y (L)* Y (L)) \\ &= \text{SQRT} (\text{PSD} (L)) \end{aligned}$$

The flow chart for achieving this is shown in figure 4.10. Firstly, each power spectral component is multiplied by a scalar factor (1.68) so that the final result will be compatible with the attenuator setting of the D/A converter. The routine for finding the square root of a number is based on the fact that any perfect square integer is equal to the sum of a series of odd integers.

ie.

$$\begin{aligned} 4 &= 2*2 = 1 + 3 \\ 9 &= 3*3 = 1 + 3 + 5 \\ 16 &= 4*4 = 1 + 3 + 5 + 7 \\ 25 &= 5*5 = 1 + 3 + 5 + 7 + 9 \\ &\text{etc.} \end{aligned}$$

The actual method for exploiting this law is commonly known and is presented in texts such as reference [1] and is adapted in this case to binary arithmetic.

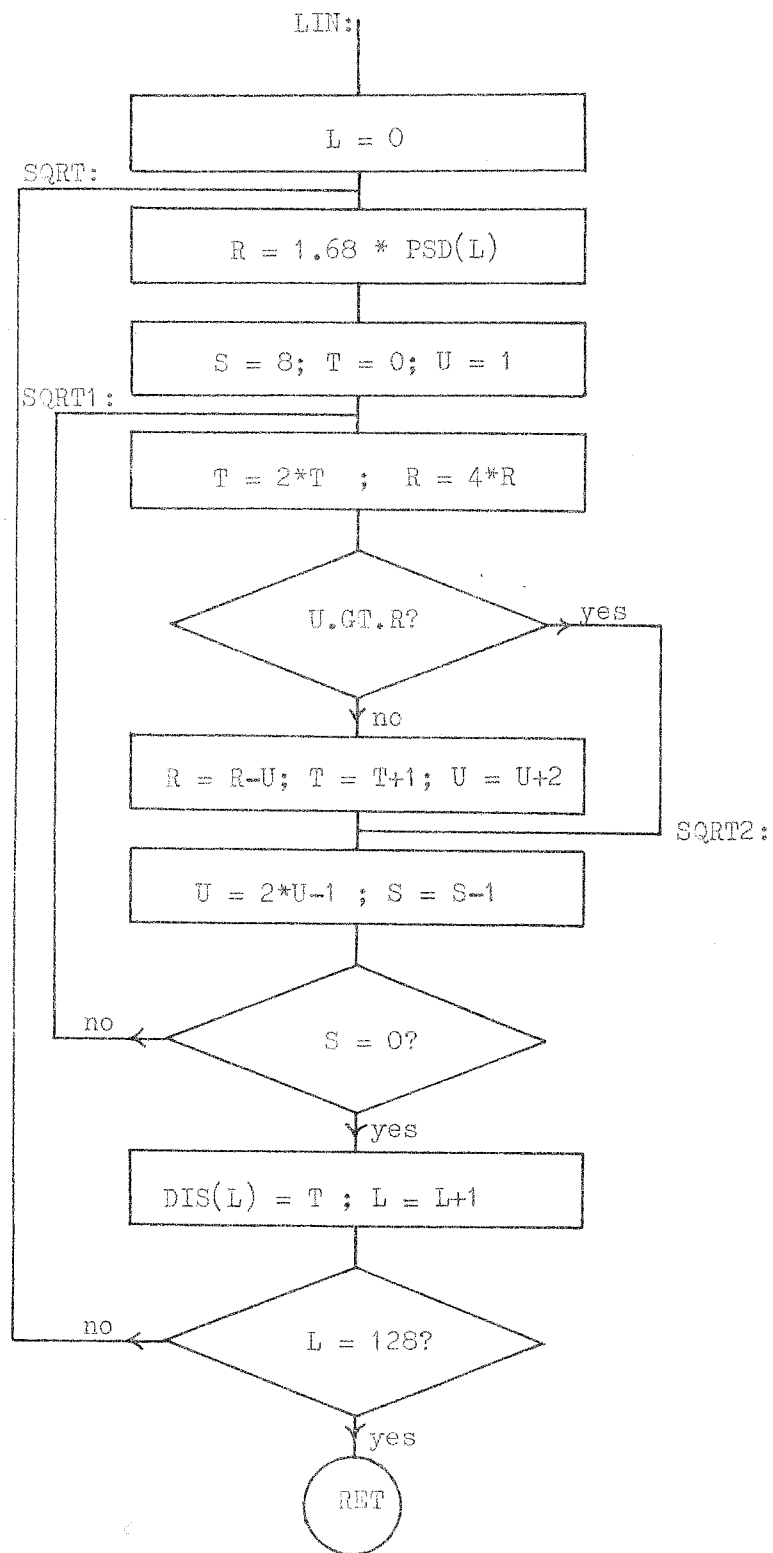


Figure 4.10 : Calculation of Square Root  
of Power Spectral Density

DISPLAY OF SPECTRUM ON A CATHODE RAY OSCILLOSCOPE

The flow chart for this routine is shown in figure 4.11. It assumes that the spectrum analyser is connected to a dual beam CRO with one Y input connected to the 'CRO Output' connector and the second Y input connected to the 'trigger output' connector. The CRO is set to trigger on the negative transition of the second input. The positive edge of the second trace coincides with the frequency which is displayed on the front panel of the analyser. This positive edge is referred to as the 'cursor' and its horizontal position on the screen can be controlled by a 'cursor control' switch on the front panel. This facility enables easy identification of spectral components by moving the cursor to coincide with a particular component and then reading the frequency from the display. The shape of the spectrum is displayed on the Y 1 input continuously for 1/8th of a second by this routine. Near the end of the main program this routine is called repeatedly until the recorder is selected. This enables the cursor to be moved at the rate of 8 components/second. Illustration of its use can be seen in the chapter 'Results'.

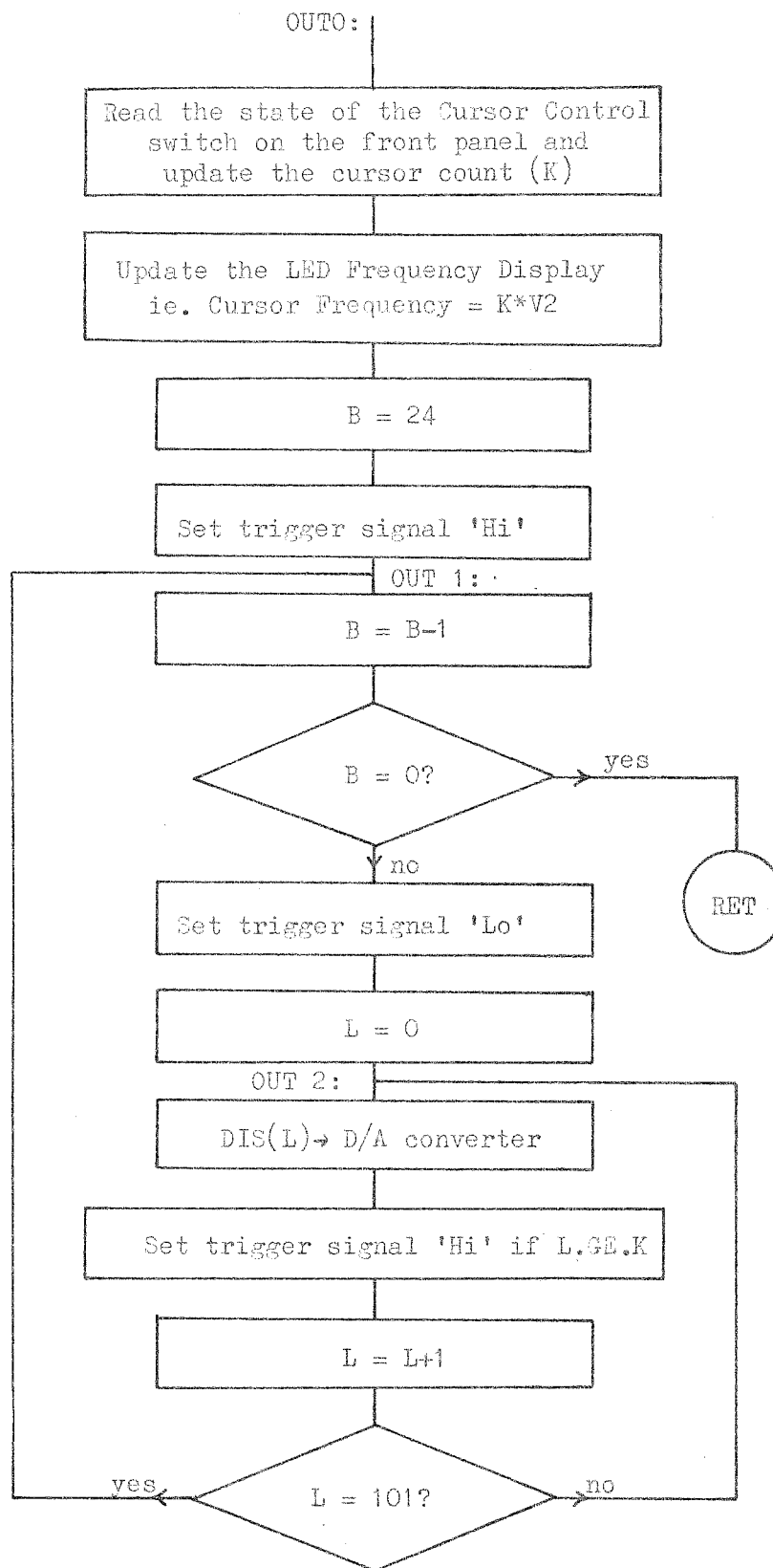


Figure 4.11 : Display of Spectrum on a Cathode Ray Oscilloscope

4.13

DISPLAY OF SPECTRUM ON A PAPER RECORDER

The flow chart for this routine is shown in figure 4.12. This facility allows a permanent record to be made of a particular spectrum. The recorder used had a fairly slow pen movement and paper speed and so this routine is time consuming ie. the output rate is one spectral point/sec which gives 100 seconds for a complete record. To facilitate reading the record a marker pen is energized at every tenth spectral point.



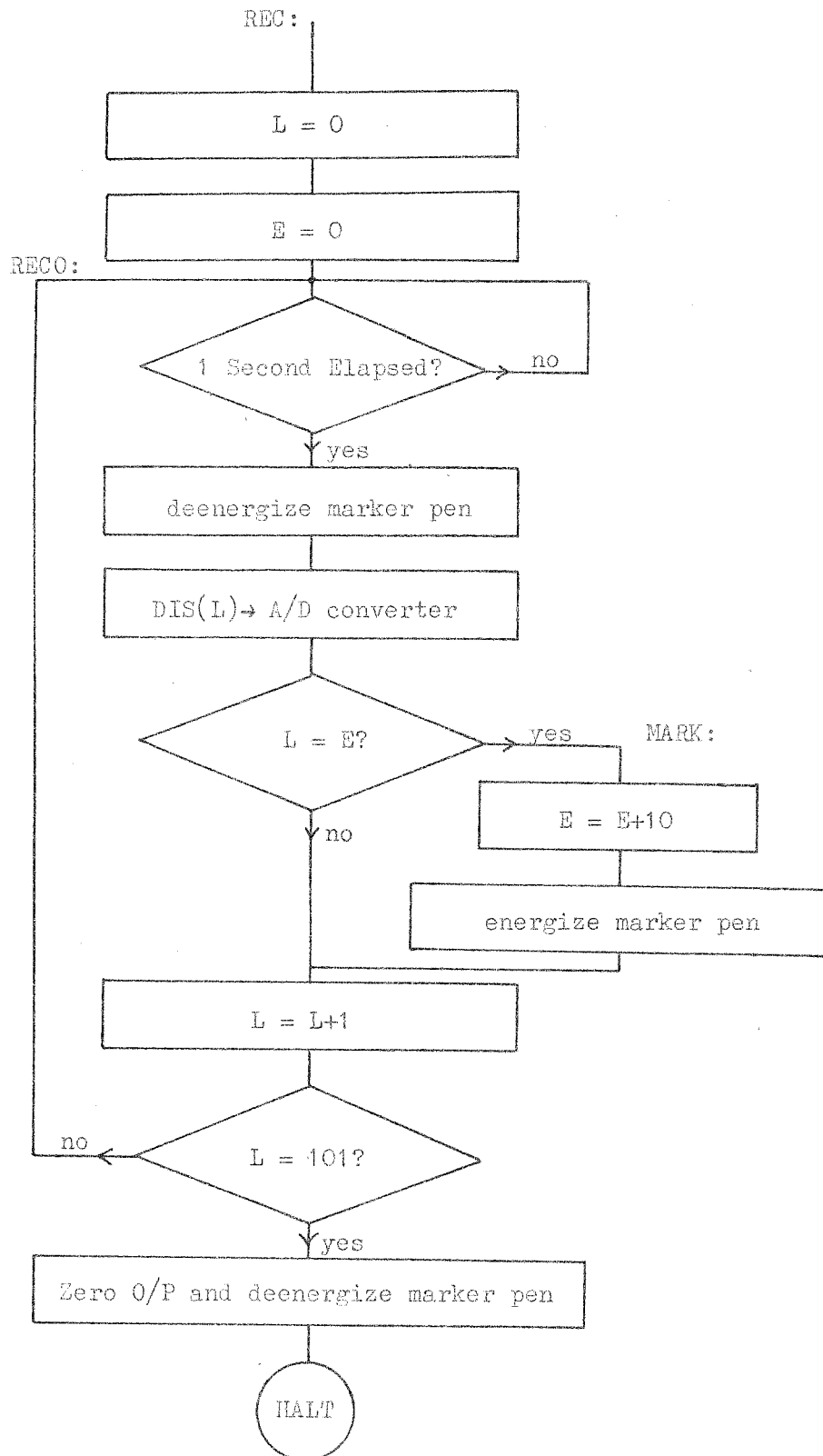


Figure 4.12 : Display of Spectrum on a Paper Recorder

REFERENCE

- [1] pp 59 - 62 of "Understanding Digital Computers"  
by Paul Siegel, published by Wiley and Sons.
- [2] "INTEL 8080 Assembly Language Programming Manual"  
INTEL Corporation, Santa Clara, USA, 1976.
- [3] "Instructions for running the INTEL 8080 Cross-Assembler,  
Cross-Compiler and Simulator"  
Local Publication 411, Computing Centre,  
University of Adelaide, December 1977.
- [4] "INTERP/80 User's Manual"  
INTEL Corporation, Santa Clara, USA 1975.

RESULTS

In this chapter the results are presented in the form of photographs of traces exhibited on a CRO. The CRO is triggered by the top trace and where applicable the positive edge of that trace represents the cursor.

In figures 5.1 to 5.9 the signal being examined is a 27 Hz sinusoid. The bandwidth setting of the analyser is set to 100 Hz. For this setting the input signal is sampled 256 times for each analysis at a rate of 256 samples/second. Hence the total sample duration = 1 second and so the spectral output samples are separated by 1 Hz. Thus 27 Hz corresponds exactly to one of the output spectral samples.

Figure 5.1 was produced by consecutively outputting the sampled sinusoid to a D/A converter. Figure 5.2 is the spectrum of the sampled signal on a linear scale. It consists of a single spectral line corresponding to 27 Hz which could also be seen at the time on the frequency display because the cursor has been aligned with this component. The scale is approximately .4 volt/cm. Figure 5.3 is the same spectrum but using a logarithmic scale which highlights the smaller components which were not apparent on the linear scale. The scale is approximately 20 db/cm. Several features require explanation.

There is a component at the beginning of the spectral record which corresponds to zero frequency. This is due to a d.c. component introduced by the offset voltage of the A/D converter and associated amplifiers.

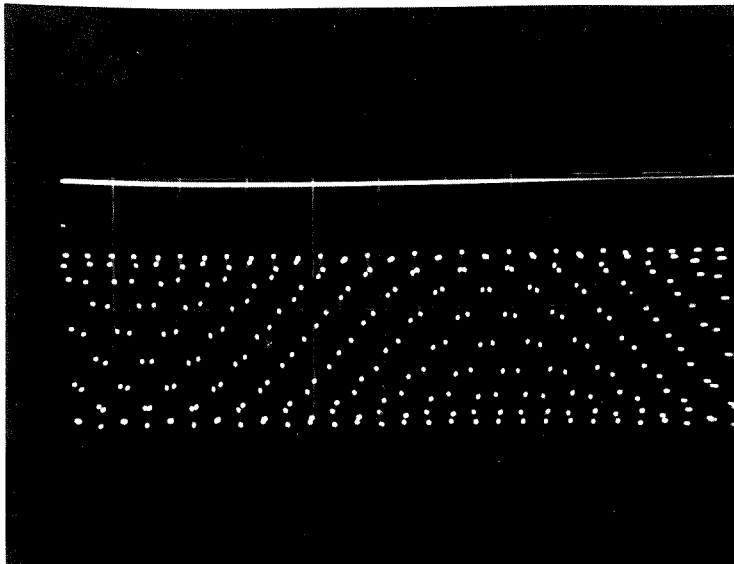


Figure 5.1:

27Hz sinusoid  
sampled at  
256 samples/sec.  
No windowing.

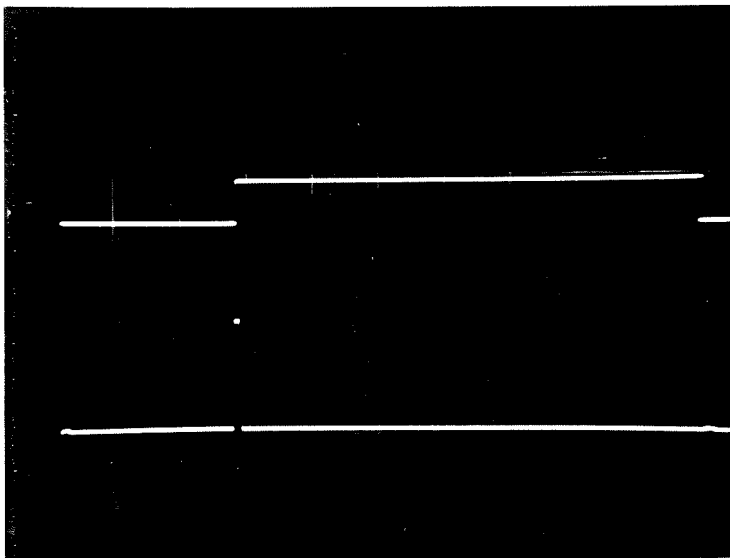


Figure 5.2:

Spectrum of  
sampled signal  
on a linear  
scale

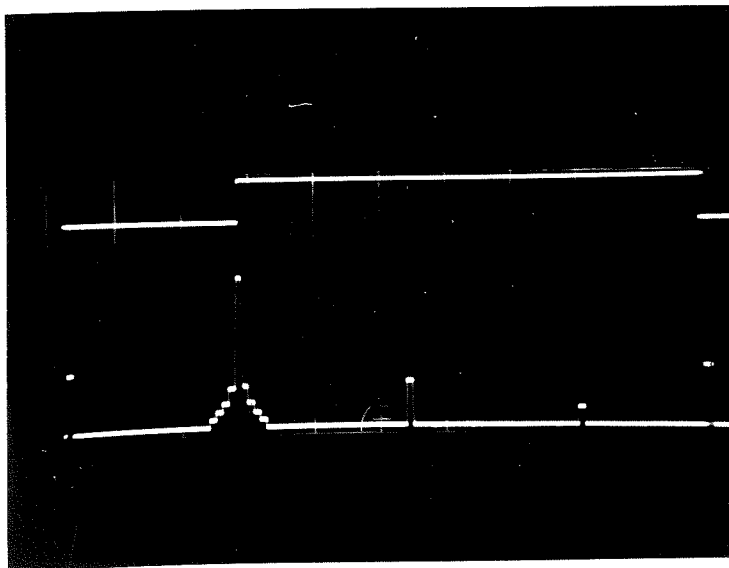


Figure 5.3:

Spectrum of  
sampled signal  
on a logarithmic  
scale

This same component can also be seen on the far right of the screen where the spectrum is repeating itself. The two components to the right of the 27 Hz component are the 2nd and 3rd harmonics of the fundamental. Another interesting feature is the occurrence of side-lobes about the fundamental. These are small compared with later records but can be made to disappear entirely by carefully adjusting the input frequency to coincide exactly with an output component.

Figure 5.4 shows a sampled 27 Hz sinusoid with the samples at the beginning and end of the record attenuated by multiplying the input array by the modified Bingham Window. Figures 5.5 and 5.6 are the spectrum of this signal on linear and logarithmic scales respectively. The features of this spectrum are similar to 5.2 and 5.3 except that there are larger side-lobes about the fundamental.

Figure 5.7 shows a 27 Hz sampled sinusoid operated on by a Hanning Window and figures 5.8 and 5.9 show the spectrum of this signal on linear and logarithmic scales respectively. Again the spectrum is similar to 5.2 and 5.3 except that the spectral components have been broadened. It is no longer possible to get a single spectral line corresponding to a pure sinusoid. The single line has been fattened by 2 adjacent components at half value. This effect is also evident in the d.c. component and the 2nd harmonic. There are other spurious components appearing in figure 5.9 which are approximately 50 db below the fundamental. This is probably due to a small amount of numerical noise being introduced by the Hanning Window routine.

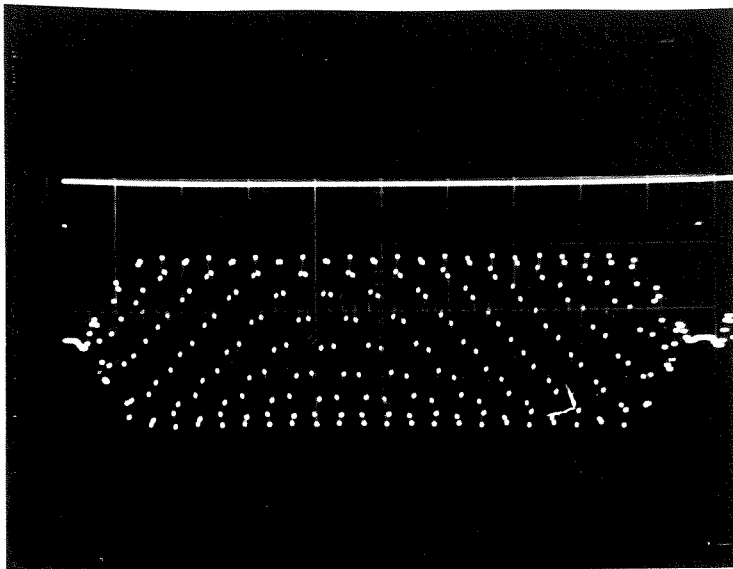


Figure 5.4:

27Hz sinusoid,  
weighted by  
modified Bingham  
window

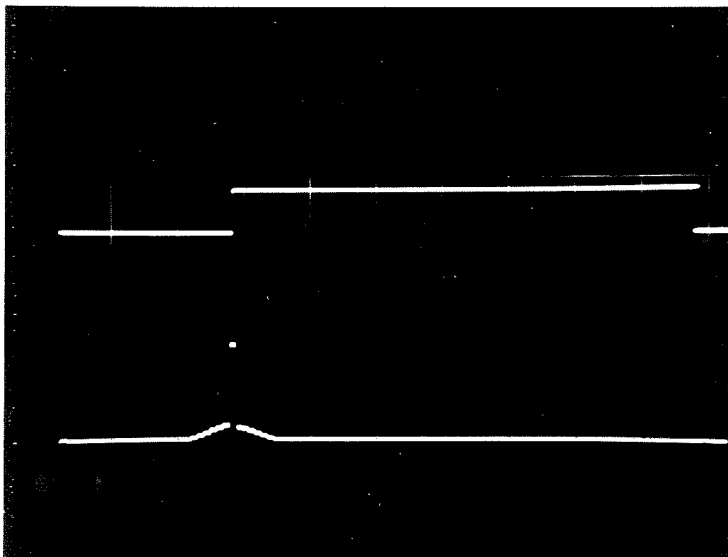


Figure 5.5:

Spectrum of  
sampled signal  
on a linear  
scale

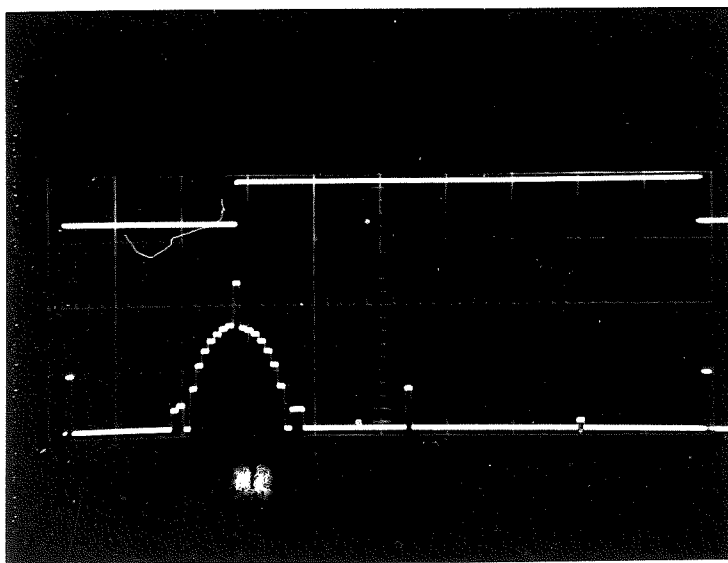


Figure 5.6:

Spectrum of  
sampled signal  
on a logarithmic  
scale

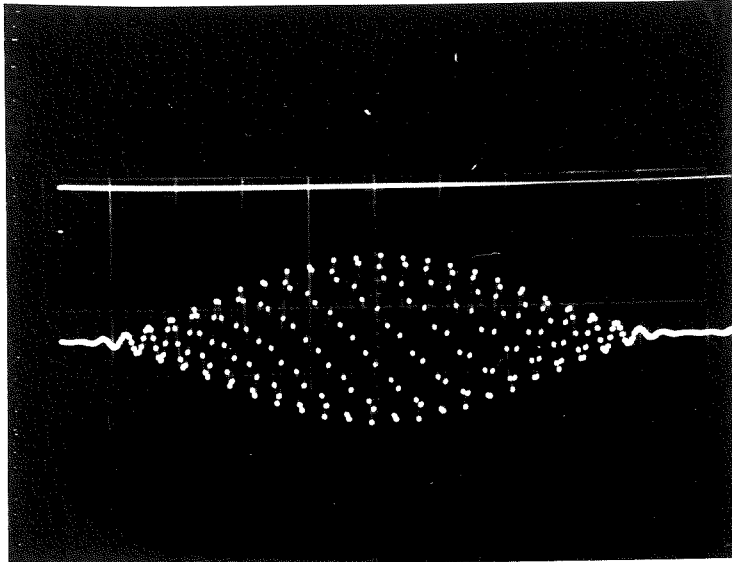


Figure 5.7:

27Hz sinusoid  
weighted by  
Hanning window

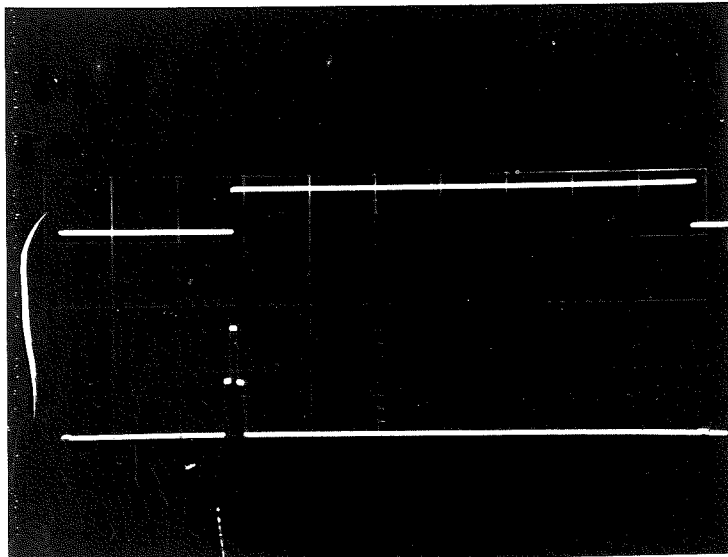


Figure 5.8:

Spectrum of  
sampled signal  
on a linear  
scale

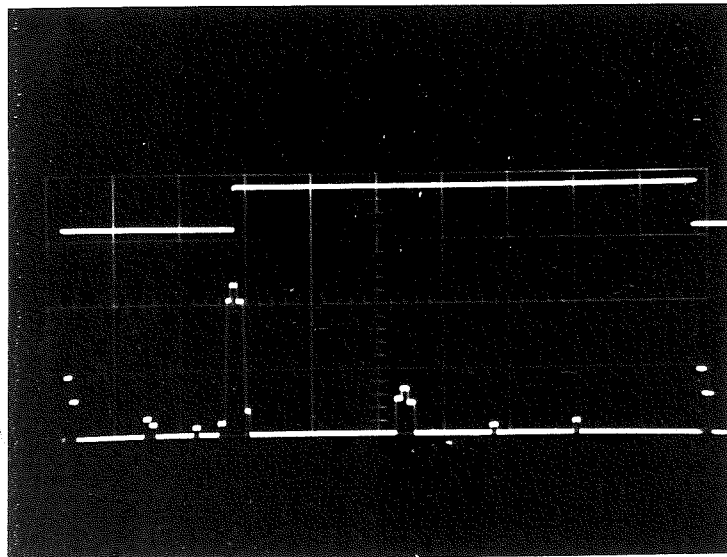


Figure 5.9:

Spectrum of  
sampled signal  
on a logarithmic  
scale

At this stage it may appear that windowing the input signal offers no advantage but this is only the case when the input signal has its frequency components corresponding very closely to the output frequency samples.

In figures 5.10 to 5.15 the spectra of a sinusoid are again shown but the frequency has been adjusted to approximately 26.5 Hz which is half-way between 2 output spectral samples (ie. 26 Hz and 27 Hz).

Figures 5.10 and 5.11 show the spectrum of the 26.5 Hz sinusoid on linear and logarithmic scales respectively with no windowing applied. The fundamental is represented by 2 approximately equal adjacent components. However the side-lobes associated with the fundamental are now so large that they nearly obscure entirely the presence of the d.c. component and the harmonics.

Figures 5.12 and 5.13 show the spectrum of the 26.5 Hz sinusoid with the modified Bingham Window applied. The fundamental is again represented by 2 approximately equal adjacent components. However the side-lobes are reduced sufficiently so that the d.c. component or any harmonic content are clearly evident.

The spectrum of the 26.5 Hz signal is repeated in figures 5.14 and 5.15 but this time a Hanning Window is used. The reduction in side-lobes is further improved. From figures 5.1 to 5.15 it can therefore be seen that, although in some cases a 'superior' spectrum can be achieved by not using any windowing, in general a more consistent and desirable spectrum is obtained by using an appropriate window function.



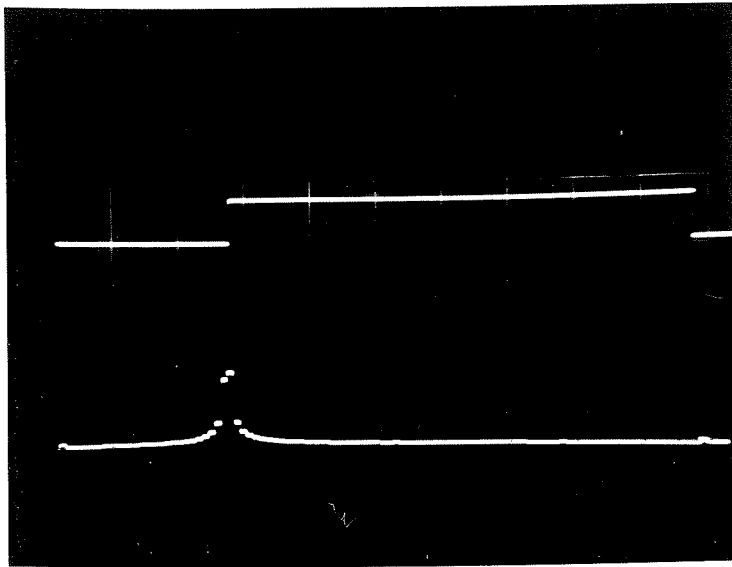


Figure 5.10: Linear spectrum of a 26.5Hz sinusoid with no windowing

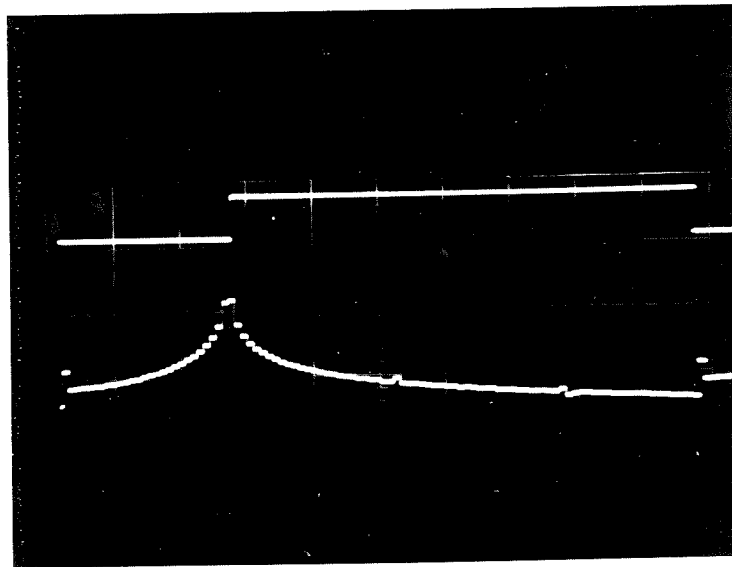


Figure 5.11: Logarithmic spectrum of a 26.5Hz signal with no windowing

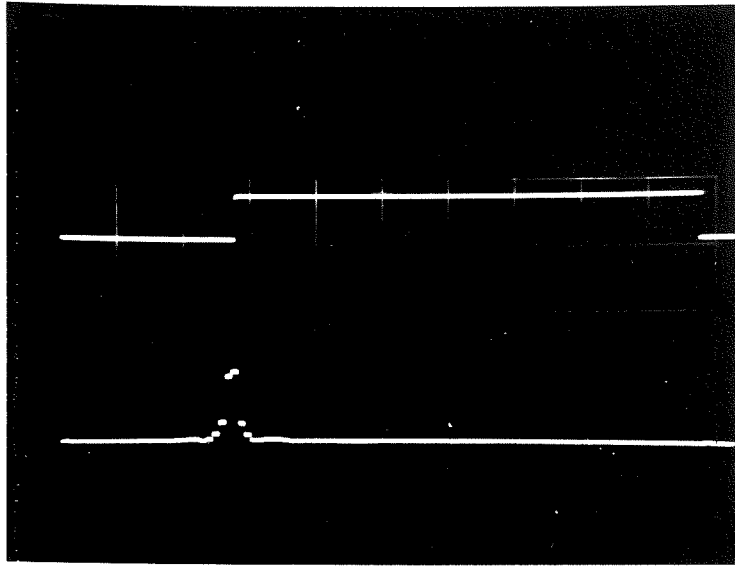


Figure 5.12: Linear spectrum of a 26.5Hz sinusoid weighted by a modified Bingham window

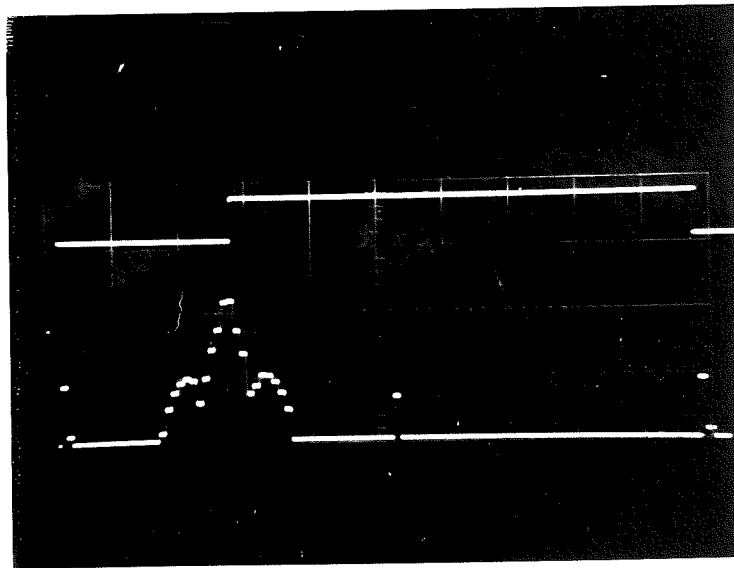


Figure 5.13: Logarithmic spectrum of a 26.5Hz sinusoid weighted by a modified Bingham window

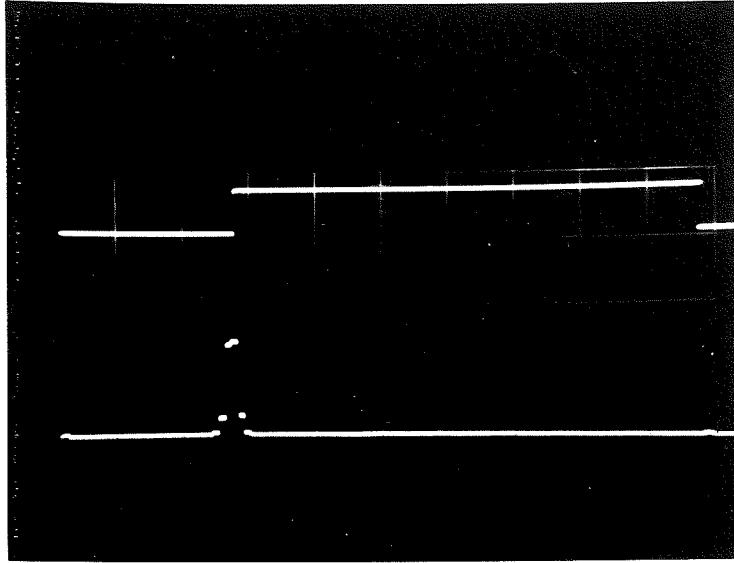


Figure 5.14: Linear spectrum of a 26.5Hz sinusoid weighted by a Hanning window

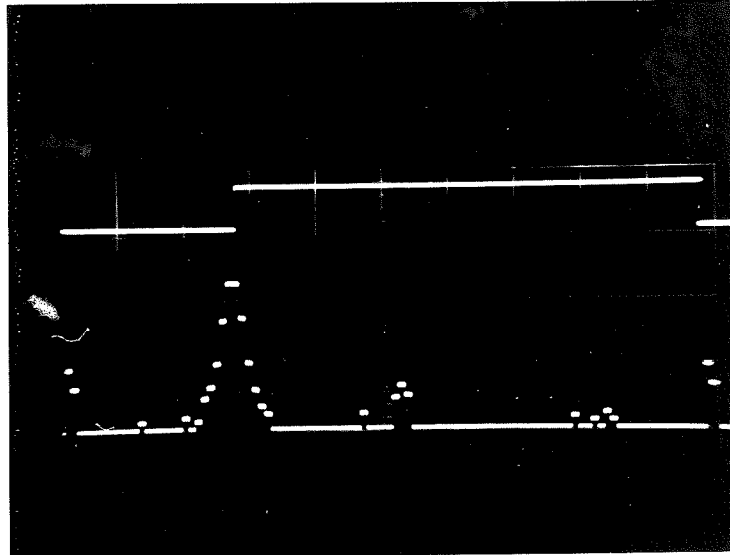


Figure 5.15: Logarithmic spectrum of a 26.5Hz sinusoid weighted by a Hanning window

Figure 5.16 shows a 27 Hz square wave sampled at 256 samples/second. The slight overshoot seen on the waveform is due to the step response of the elliptic low pass input filters. Figures 5.17 and 5.18 show the spectrum of this signal on a linear and logarithmic amplitude scale respectively. A perfect square wave has only odd harmonics whereas a 2nd harmonic is apparent. This is probably due to the poor waveshape produced by this particular oscillator.

Figures 5.19 to 5.21 show the sampled square wave operated on by a modified Bingham window and its spectrum and this is again repeated in figures 5.22 to 5.24 using a Hanning Window. The same features appear as for a sinusoid except that there are larger harmonics.

Figure 5.25 shows the spectrum on a logarithmic amplitude scale of a 27 Hz square wave with no window employed and the sample rate doubled. Since the sampling time is halved the output frequency components are even integers ie. 0, 2, 4, ..., 200 Hz. Hence the 27 Hz fundamental and the odd harmonics are half-way between output frequency components and so exhibit large side-lobes whereas the even harmonics have negligible side-lobes due to the fact that they each correspond closely to an output frequency component.

Figure 5.26 is included to show simply that a waveform is more easily 'recognizable' when it is sampled at a rate much greater than its fundamental frequency component.

Figures 5.27 to 5.29 are given to illustrate the veracity of the FFT routine. A test input array is set up which has the properties :

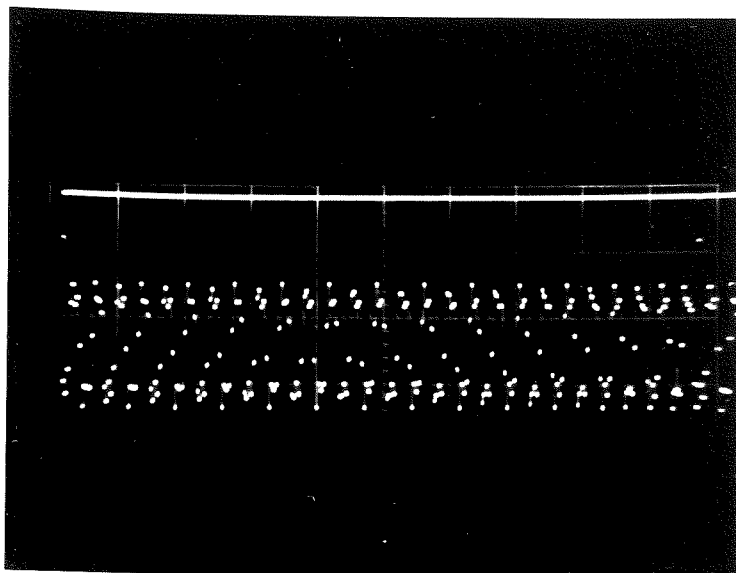


Figure 5.16:

27Hz square wave  
sampled at  
256 samples/sec.  
No windowing.

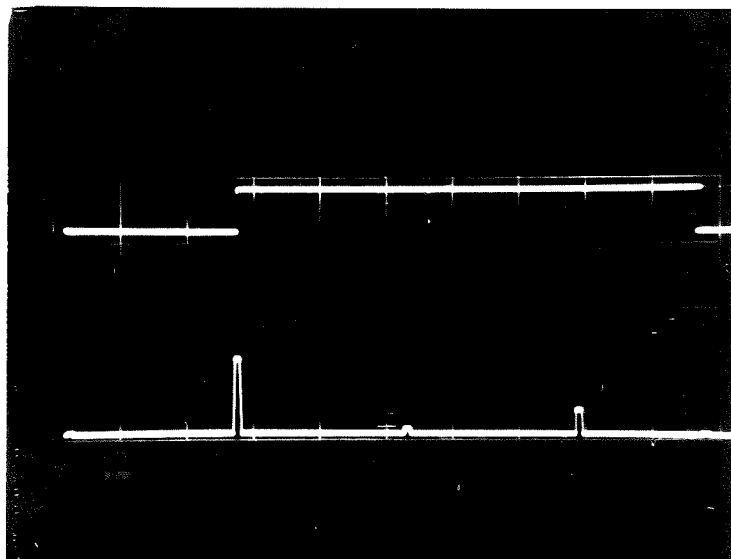


Figure 5.17:

Spectrum of  
sampled signal  
on a linear  
scale

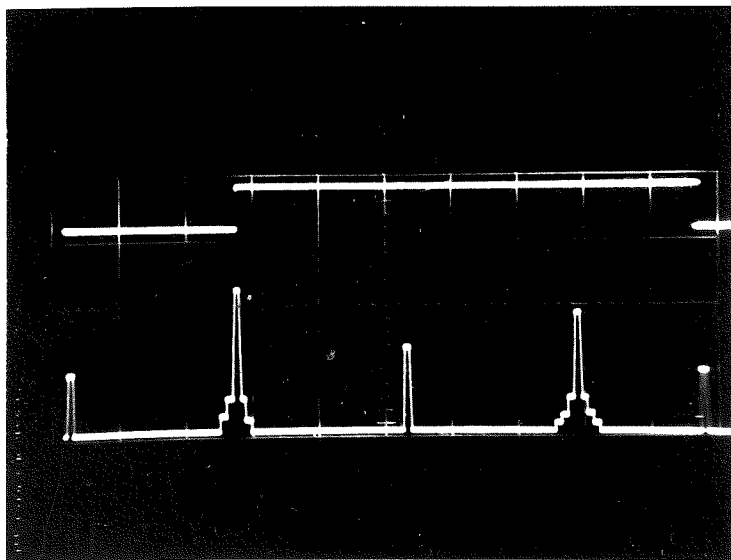


Figure 5.18:

Spectrum of  
sampled signal  
on a logarithmic  
scale

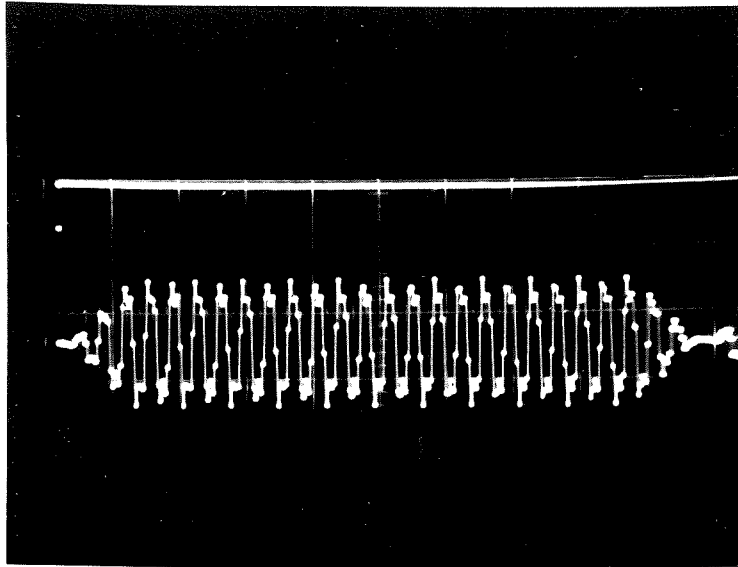


Figure 5.19:

27Hz square wave  
weighted by a  
modified Bingham  
window

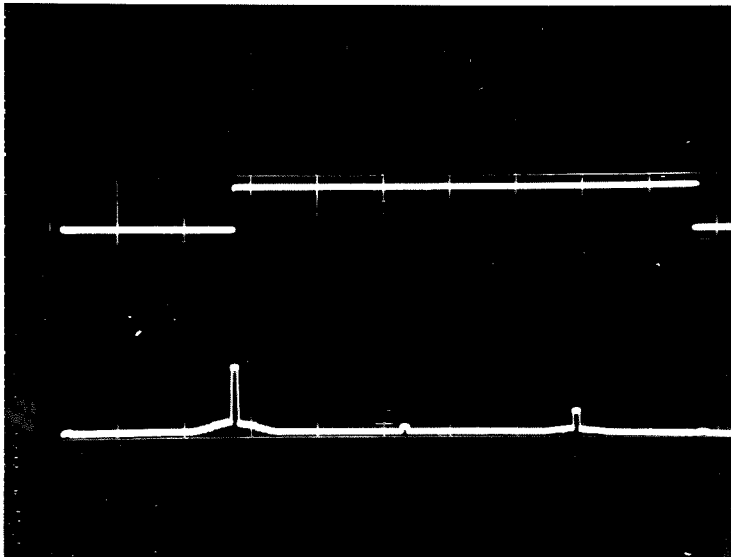


Figure 5.20:

Spectrum of  
sampled signal  
on a linear  
scale

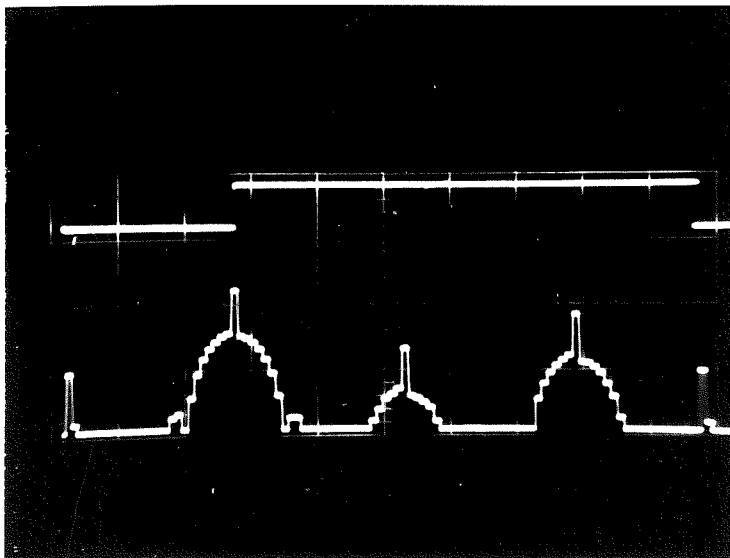


Figure 5.21:

Spectrum of  
sampled signal  
on a logarithmic  
scale

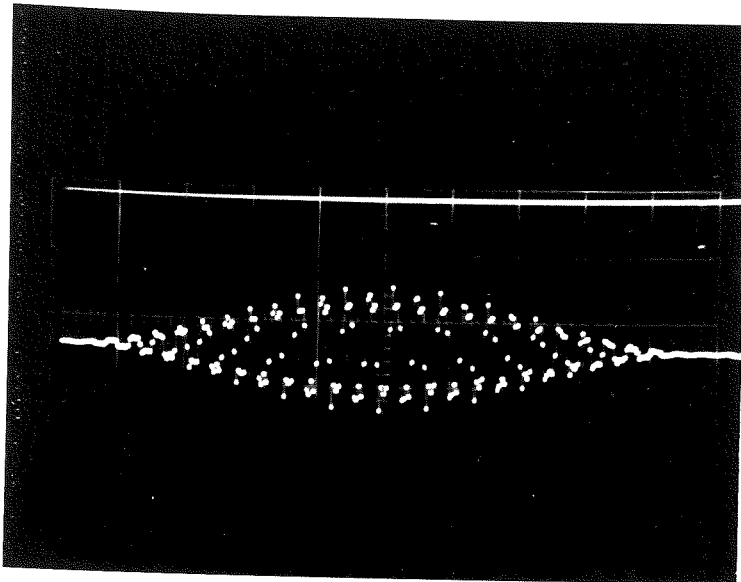


Figure 5.22:

27Hz square wave  
modified by a  
Hanning window

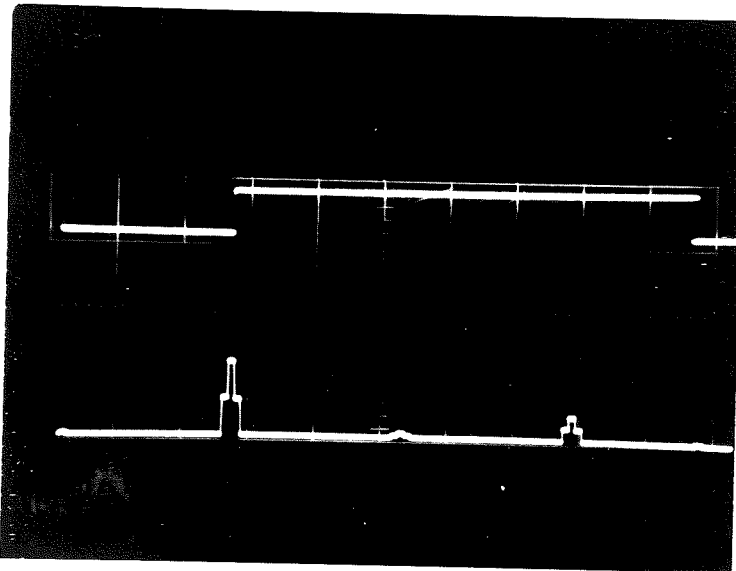


Figure 5.23:

Spectrum of  
sampled signal  
on a linear  
scale

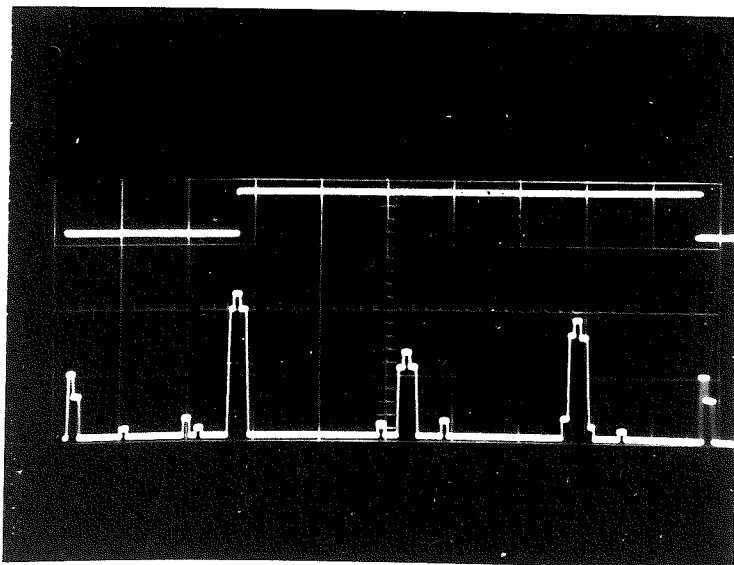


Figure 5.24:

Spectrum of  
sampled signal  
on a logarithmic  
scale

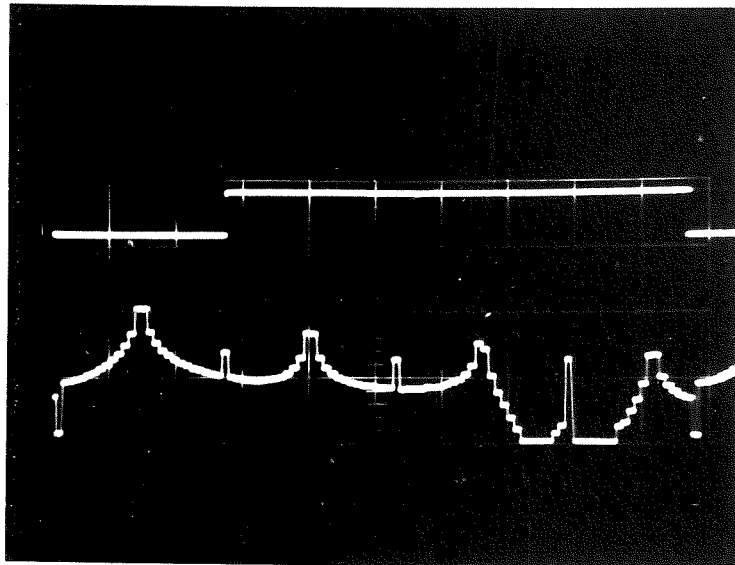


Figure 5.25: Spectrum of a 27Hz square wave sampled at 512 samples/sec. Logarithmic scale.

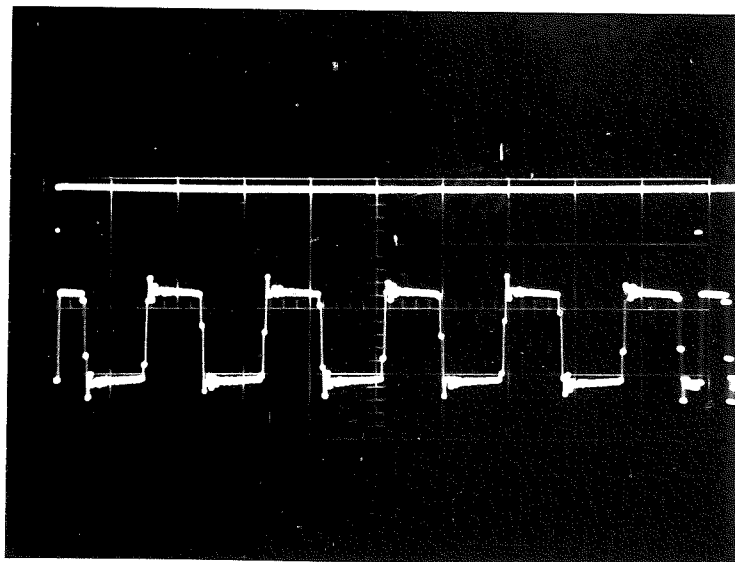


Figure 5.26: 27Hz square wave sampled at 1.28kHz



$$X_k = \begin{cases} 1 & \text{for } k = 0 \dots (N/2-1) \\ 0 & \text{for } k = N/2 \dots (N-1) \end{cases}$$

In Appendix 3 it is shown that :

$$X_0 = 0.5$$

$$X_n = 0 \quad \text{for } n \text{ even}$$

$$X_n = (1 - j \cos(\Pi n/N) / \sin(\Pi n/N)) / N \quad \text{for } n \text{ odd}$$

Figure 5.27 shows the input array  $X_k$  and figures 5.28 and 5.29 show the imaginary part of  $X_n$

$$\text{Im}(X_n) = \begin{cases} 0 & \text{for } n = 0 \\ 0 & \text{for } n \text{ even} \\ -(1/N) \cdot \cos(\Pi n/N) / \sin(\Pi n/N) & \text{for } n \text{ odd} \end{cases}$$

$$\text{For } n \ll N/2$$

$$\text{Im}(X_n) \propto 1/n$$

From figure 5.28 it can be seen that only odd frequency components occur and that their amplitudes successively decrease at a  $1/n$  rate.

$$\begin{aligned} \text{Im}(X_{N-n}) &= - (1/N) \cos(\Pi(N-n)/N) / \sin(\Pi(N-n)/N) \\ &= (1/N) \cos(\Pi n/N) / \sin(\Pi n/N) \\ &= - \text{Im}(X_n) \end{aligned}$$

ie. the imaginary part of  $X_n$  is odd about the  $n = N/2$  axis which can be seen from figure 5.29.

Figures 5.30 to 5.34 are included to illustrate the effect of spectral averaging. The input signal consists of some constant frequency components which are mixed with noise. In figure 5.30 the spectrum is fairly meaningless but as the number of accumulations increases the nature of the spectral content of the input signal can be ascertained more accurately. From figure 5.34, which is obtained by averaging 64 spectra, it can be seen that there are several fixed frequency components amidst a background of noise which is substantially 'white' over the measured frequency range.

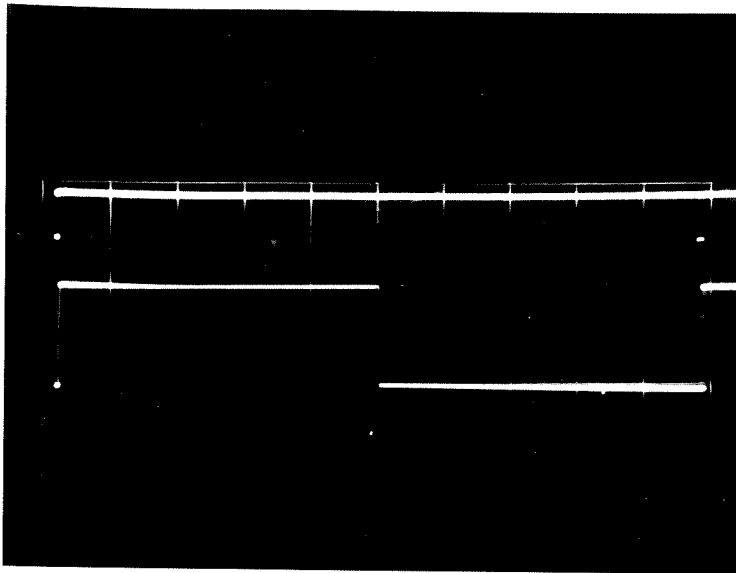


Figure 5.27:

Step function

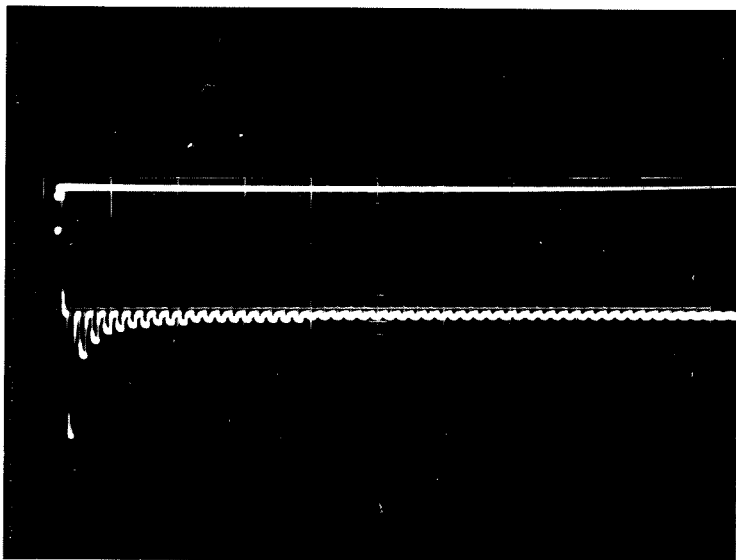


Figure 5.28:

Imaginary part  
of DFT of step  
function

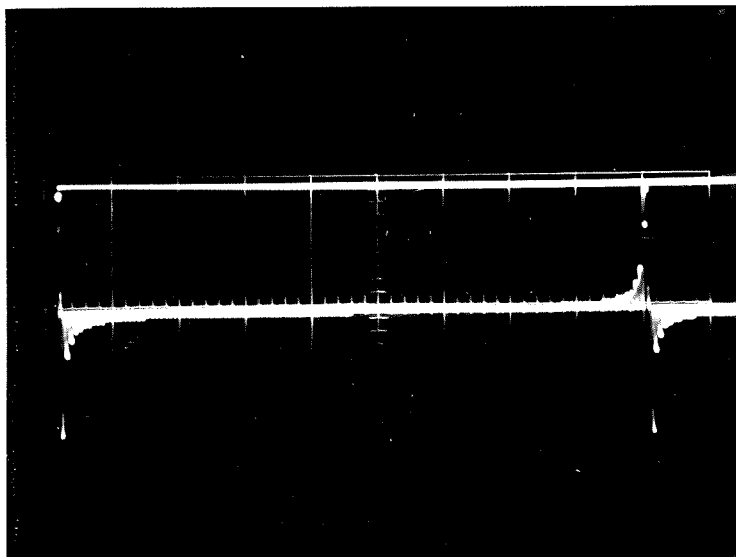


Figure 5.29:

Imaginary part  
of DFT on an  
expanded scale

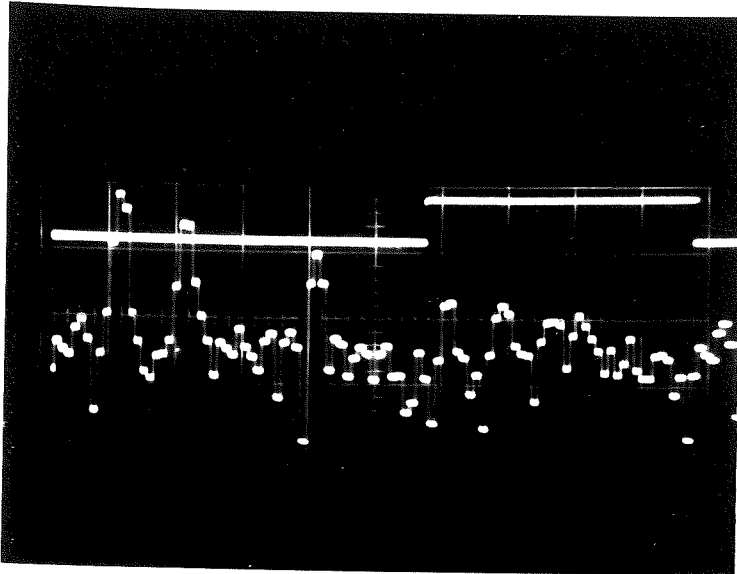


Figure 5.30:

Signal plus noise  
one accumulation  
logarithmic scale

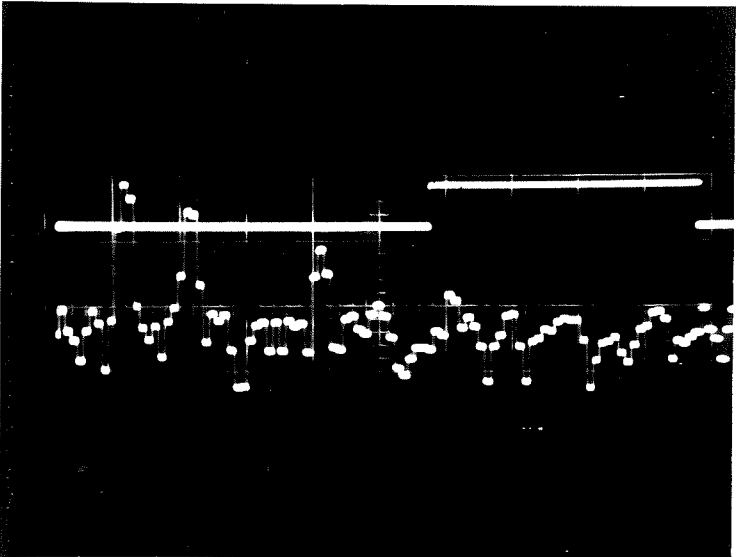


Figure 5.31:

Two accumulations

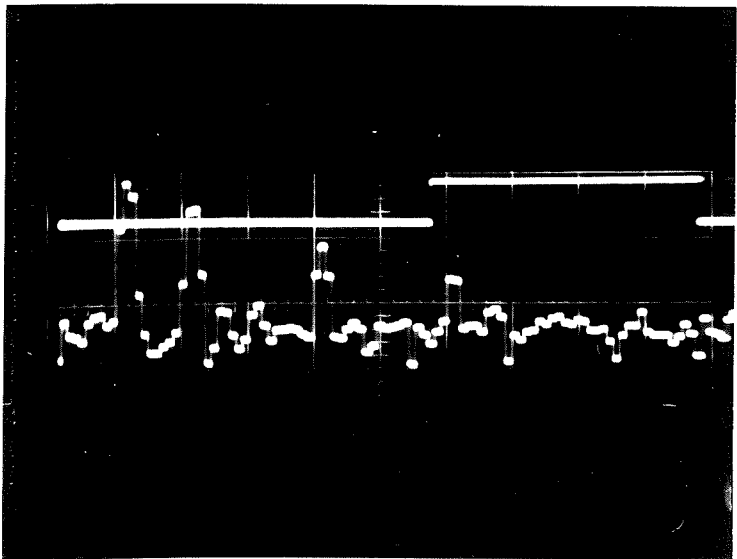


Figure 5.32:

Four  
accumulations

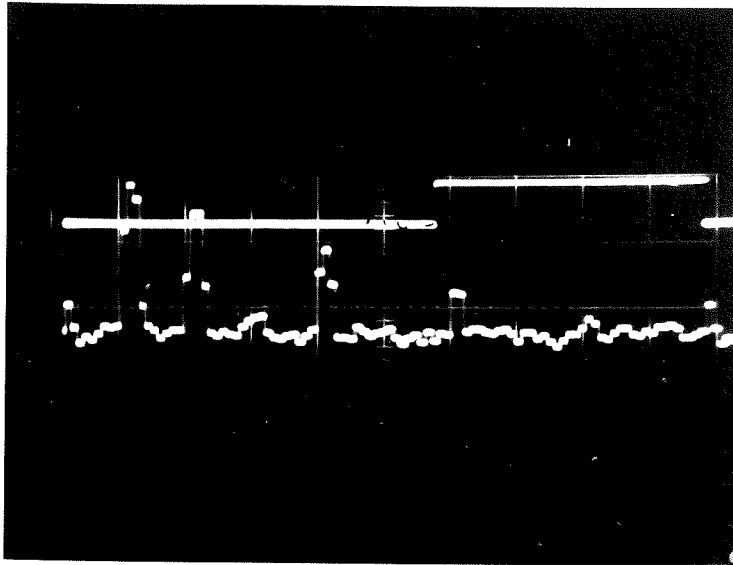


Figure 5.33:

Sixteen  
accumulations

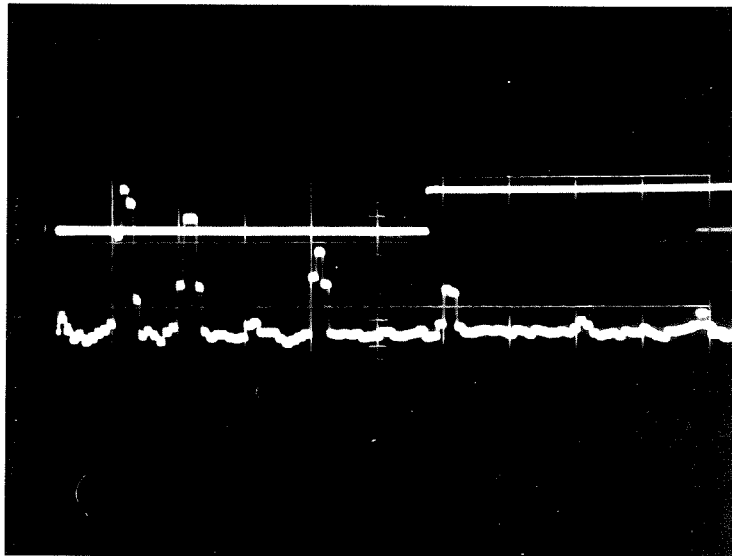


Figure 5.34:

Sixty four  
accumulations

6.

CONCLUSION

From the previous chapters it can be seen that sophisticated fourier techniques, previously only used on large computers, are now able to be incorporated in relatively cheap and small micro-computer systems. The analyser presented is fairly crude when compared with the measurement capability or extra features of an instrument such as Hewlett Packard's Model 3582A Spectrum Analyser outlined in reference [1], however its performance could be improved further with only minor hardware additions as follows :

(a) Improvement in Frequency Resolution

The present analyser performs a 256 point DFT. A 1024 point DFT could be easily implemented by adding an extra 3072 (3K) words of RAM and making only minor alterations to the existing software. This would improve resolution by a factor of 4 which would greatly enhance the instrument's usefulness.

(b) Digital Display of Amplitude

At present a digital display of frequency is provided which is used in conjunction with the cursor control. The same display circuitry could be used to display amplitude in db or millivolts by the addition of 1 switch and an extra subroutine.

(c) Flat-top Window

From figures 2.8, 2.10 and 2.12 from chapter 2 it can be seen that there is substantial variation in amplitude of the spectrum of the existing window functions over the range  $(-\frac{1}{2T}, \frac{1}{2T})$  where  $T$  is the sampling time. The implication of this is that a sinusoid with frequency in between 2 output frequency points will be reduced in amplitude on the output of the spectrum.

This is a serious limitation to the measurement of the amplitudes of the spectral components.

The problem can be overcome, however, by the use of a 'flat-top' window as described in [2]. This function has a substantially flat spectrum over the range  $(-\frac{1}{2T}, \frac{1}{2T})$  which would allow accurate amplitude measurements to be made. A suggested 'flat-top' window function is :

$$w(t) = 1.0 + 1.86 \cos 2\pi t + 0.86 \cos 4\pi t \quad -\frac{1}{2} \leq t \leq \frac{1}{2}.$$

(d) An Additional Input

If additional circuitry were added so that two input signals could be examined at the same time then the analyser could perform other useful tasks such as cross-correlation or the determination of the transfer function of an input-output port (eg. a filter).

In conclusion, there is wide scope to the applications for which an instrument such as this can be used. The limits would often be determined by the requirements of the user and the time invested in generating software.

Reference

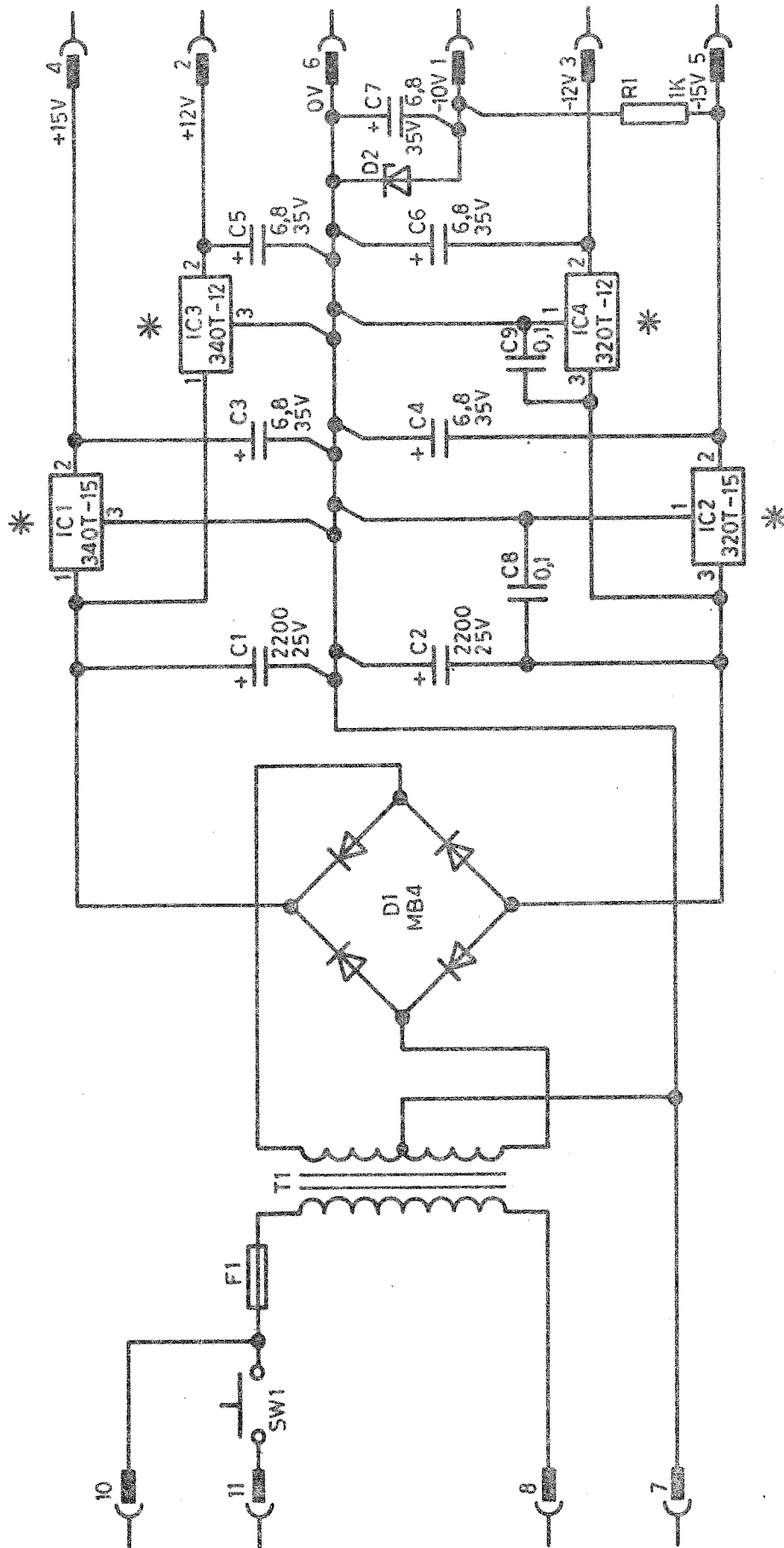
- [1] "A High-Resolution, Low-Frequency Spectrum Analyser" by Nixon A. Pendergrass and John S. Farnbach, Hewlett-Packard Journal, July 1978.
- [2] "Window Functions for Spectrum Analysis" by Roger G. Cox, Hewlett-Packard Journal, July 1978.

APPENDIX 1

DETAILED CIRCUIT DIAGRAMS







\* MOUNTED ON HEATSINKS

SPECTRUM ANALYSER

POWER SUPPLY BOARD 1

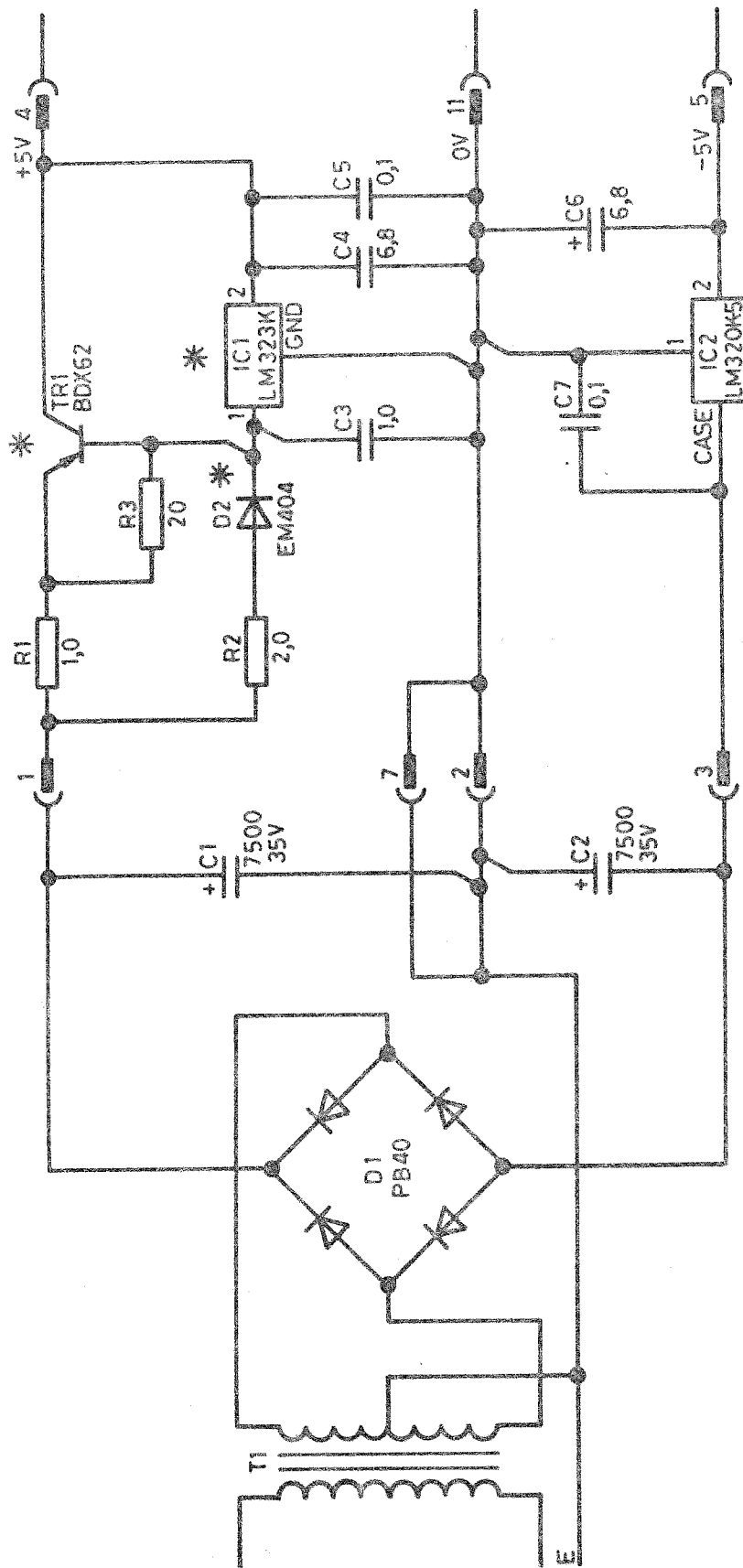
DRN	<i>L.P. Galt</i>	2/5/77
CKD	D.M. TURNER	25/5/77
INSP	<i>W.D. Galt</i>	26/5/77
AUTH	<i>J. Galt</i>	27/5/77

THE ELECTRICITY TRUST OF SOUTH AUSTRALIA

SCALE OP3

INSTRUMENTATION T.E.P

A4-T0344  
SHEET 2-3-1



CAPACITORS C3,4,5 & THE GROUND CONNECTIONS OF IC1 & IC2 TO HAVE A SINGLE POINT GROUND CONNECTION.

\* MOUNTED ON COMMON HEAT SINK

D1 & T1 ARE MOUNTED BEHIND THE SWITCH PANEL

SPECTRUM ANALYSER

POWER SUPPLY BOARD 2

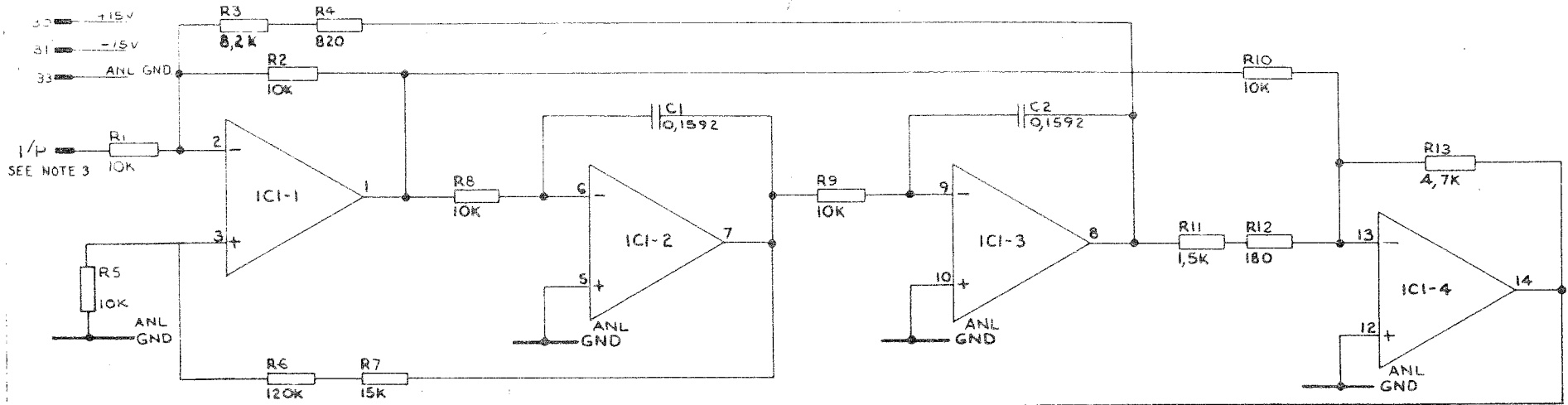
DRN	<i>B. P. R. H. T.</i>	24/1/77
CKD	D. M. TURNER	25.5.77
INSP	<i>W. G. H. S.</i>	25.5.77
AUTH	<i>J. R. H. S.</i>	25.5.77

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SCALE OP3

INSTRUMENTATION T.E.P.

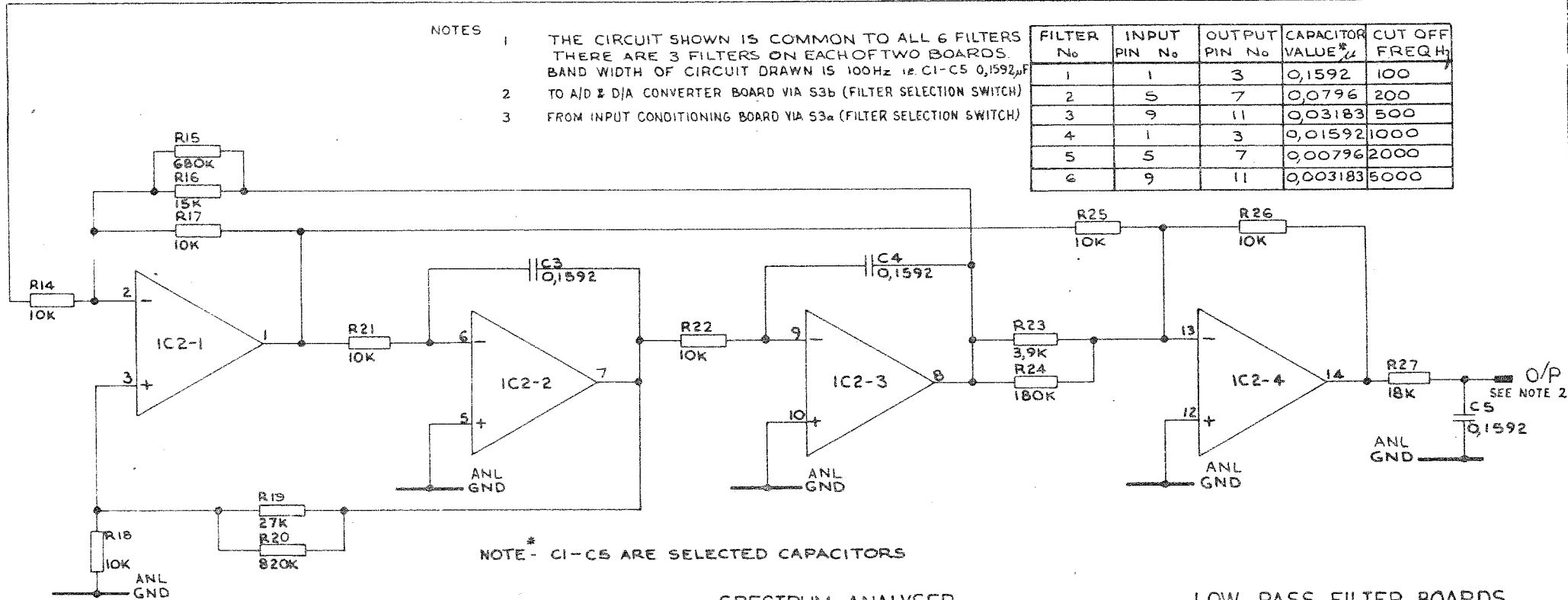
A4-T0344  
SHEET 2.4.1



NOTES

- 1 THE CIRCUIT SHOWN IS COMMON TO ALL 6 FILTERS THERE ARE 3 FILTERS ON EACH OF TWO BOARDS. BAND WIDTH OF CIRCUIT DRAWN IS 100Hz i.e. C1-C5 0,1592µF
- 2 TO A/D & D/A CONVERTER BOARD VIA S3b (FILTER SELECTION SWITCH)
- 3 FROM INPUT CONDITIONING BOARD VIA S3a (FILTER SELECTION SWITCH)

FILTER No	INPUT PIN No	OUTPUT PIN No	CAPACITOR VALUE µF	CUT OFF FREQ Hz
1	1	3	0,1592	100
2	5	7	0,0796	200
3	9	11	0,03183	500
4	1	3	0,01592	1000
5	5	7	0,00796	2000
6	9	11	0,003183	5000

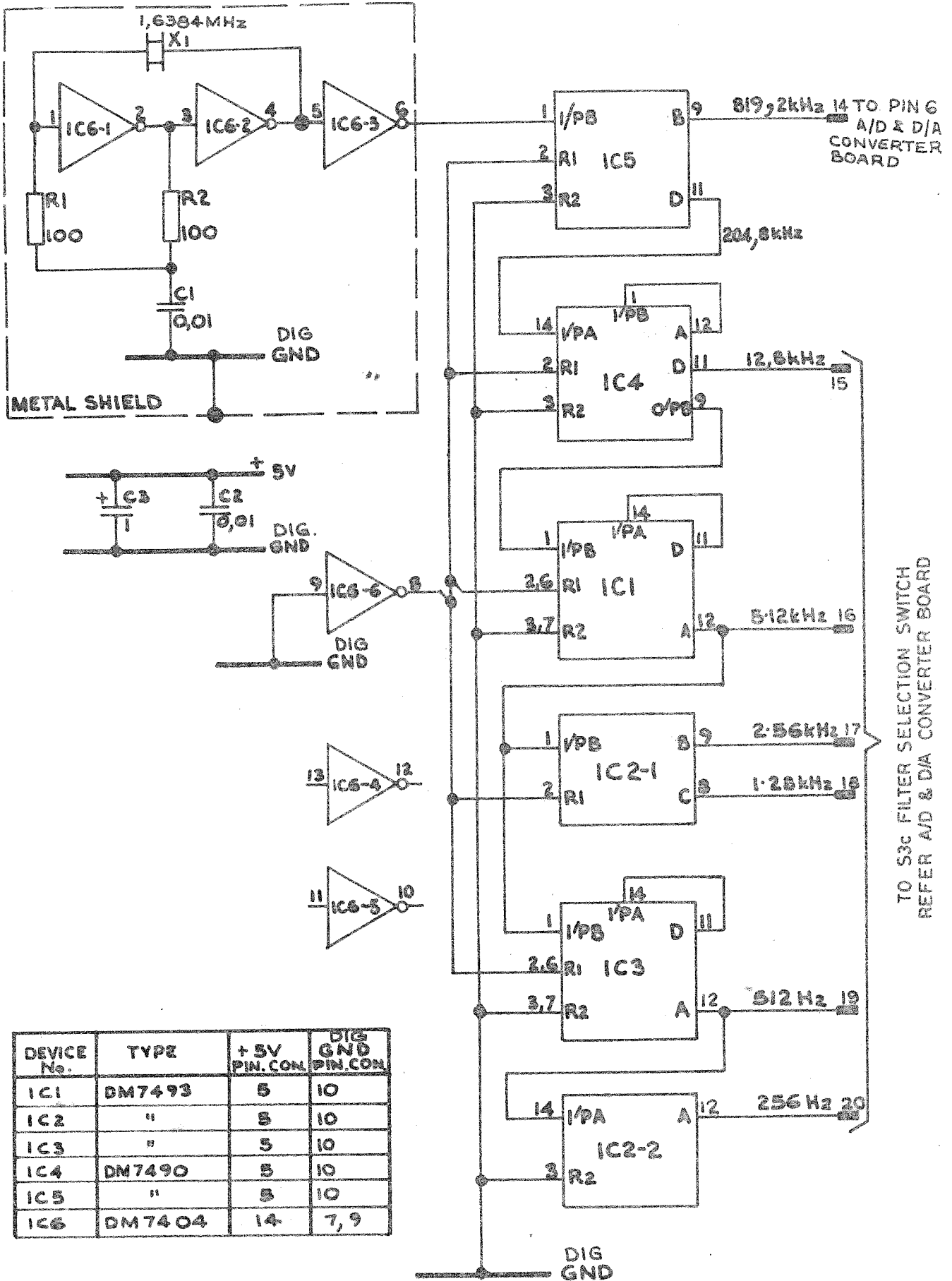


\* NOTE - C1-C5 ARE SELECTED CAPACITORS

SPECTRUM ANALYSER

LOW PASS FILTER BOARDS

REL.	DETAILS OF REVISION	REC.	REV.	APC.	DATE	DESIGNED BY D.M. TURNER 264-78	DATE 3/5/79	DR.	ELECTRICITY TRUST OF SOUTH AUSTRALIA	SCALE	DR.
						CHECKED BY <i>[Signature]</i> 255-79	255-79		INSTRUMENTATION T.E.P.		
						APPROVED BY <i>[Signature]</i> 255-79	255-79				
											A3 T0344 SHEET 2.5.1



TO S3c FILTER SELECTION SWITCH  
REFER A/D & D/A CONVERTER BOARD

DEVICE No.	TYPE	+5V PIN. CON.	DIG GND PIN. CON.
IC1	DM7493	5	10
IC2	"	5	10
IC3	"	5	10
IC4	DM7490	5	10
IC5	"	5	10
IC6	DM7404	14	7,9

SPECTRUM ANALYSER SAMPLE RATE GENERATOR

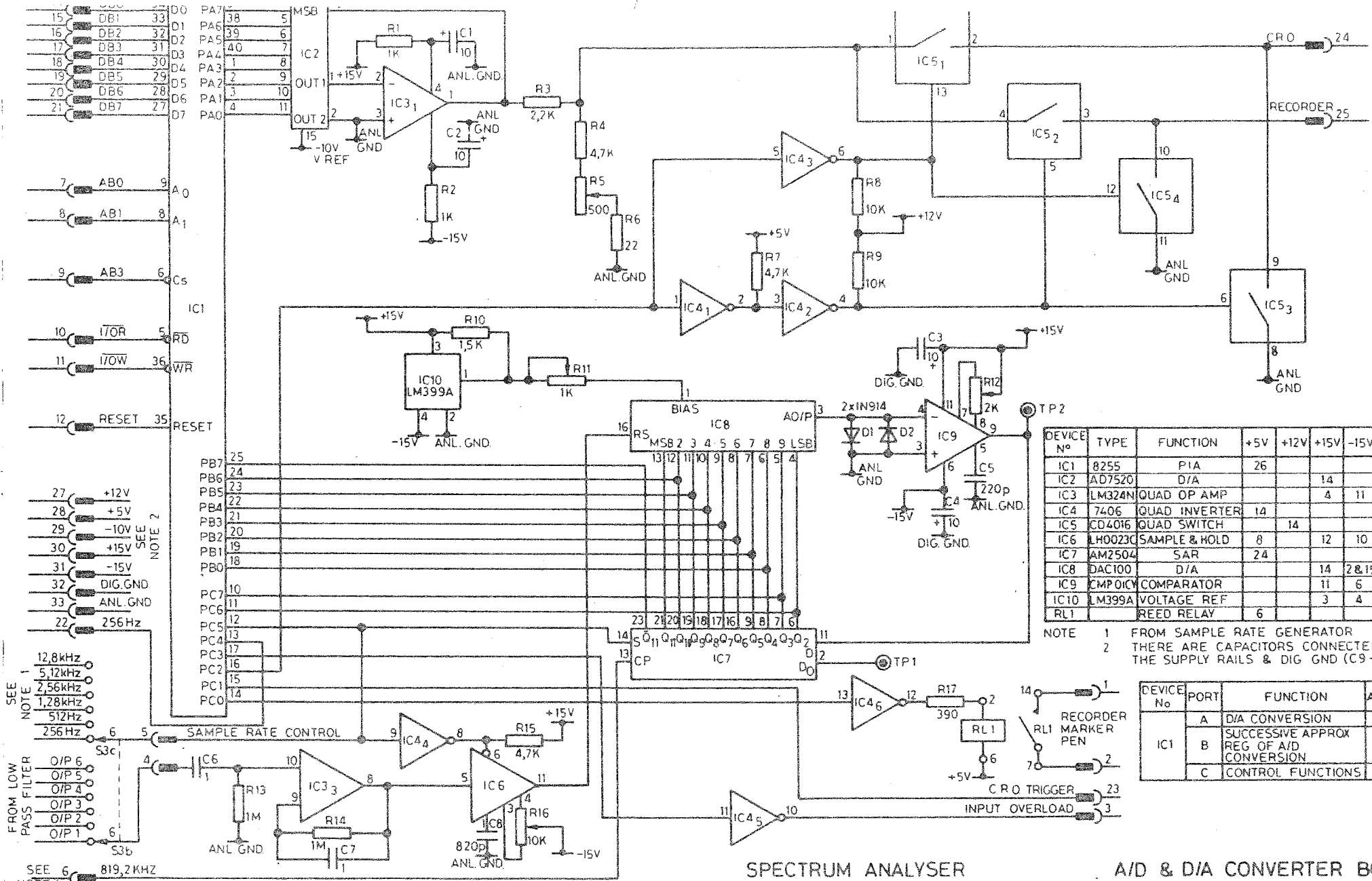
DRN	D.M.TURNER	26-47
CKD	<i>S. J. Roberts</i>	25/5/77
INSP	<i>W. Hall</i>	25/5/77
AUTH	<i>J. Hendry</i>	27.5.77

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SCALE OP3

INSTRUMENTATION T.E.P.

A4-T0344  
SHEET 2.6.1



DEVICE No	TYPE	FUNCTION	+5V	+12V	+15V	-15V	DIG GND	ANL GND
IC1	8255	PIA	26				7	
IC2	AD7520	D/A			14		3	2
IC3	LM324N	QUAD OP AMP			4	11		
IC4	7406	QUAD INVERTER	14				7	
IC5	CD4016	QUAD SWITCH		14			7	8 & 11
IC6	LH0023C	SAMPLE & HOLD	8		12	10		9
IC7	AM2504	SAR	24				1 & 12	
IC8	DAC100	D/A			14	2 & 15		
IC9	CMPOIC	COMPARATOR			11	6	2	3
IC10	LM399A	VOLTAGE REF			3	4		2
RL1		REED RELAY						

NOTE 1 FROM SAMPLE RATE GENERATOR  
 2 THERE ARE CAPACITORS CONNECTED BETWEEN THE SUPPLY RAILS & DIG GND (C9-C13)

DEVICE No	PORT	FUNCTION	ADDRESS	MODE
IC1	A	D/A CONVERSION	#F4	
	B	SUCCESSIVE APPROX REG OF A/D CONVERSION	#F5	#F7
	C	CONTROL FUNCTIONS	#F6	

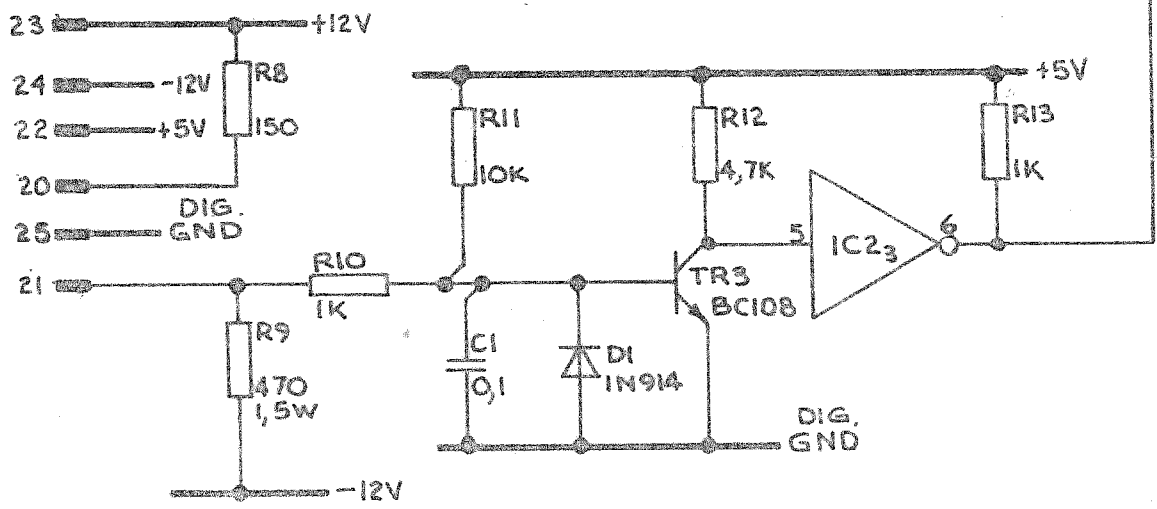
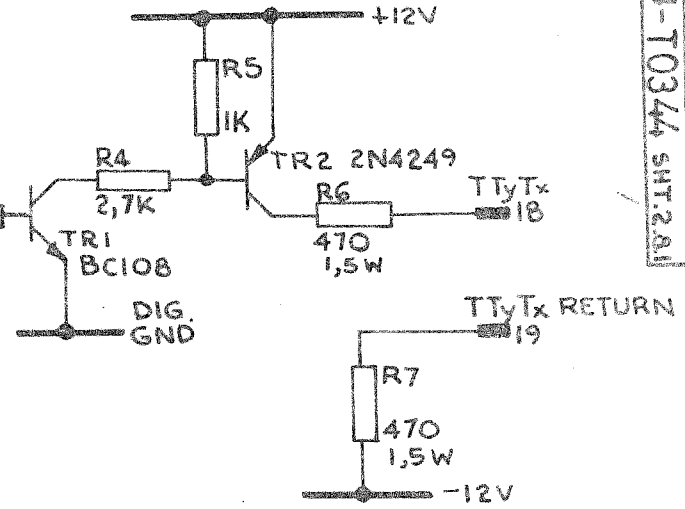
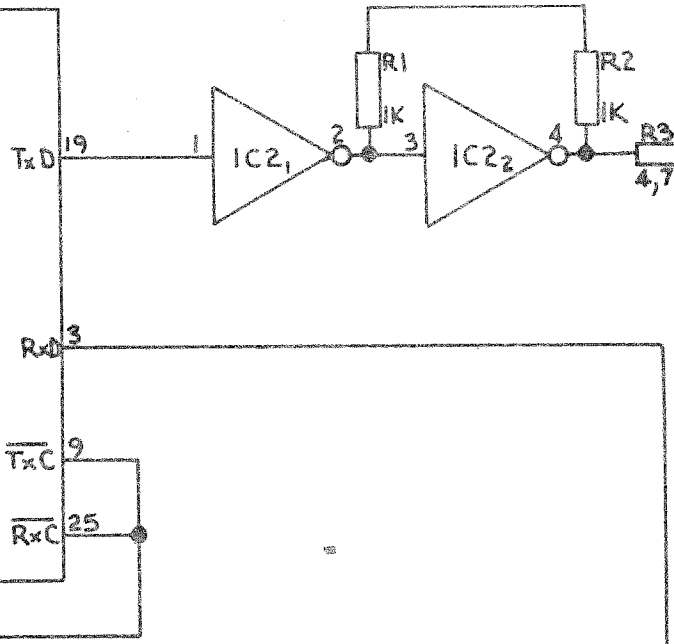
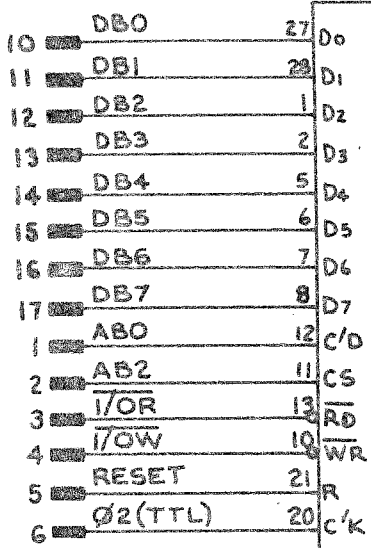
SPECTRUM ANALYSER

A/D & D/A CONVERTER BOARD

REF.	DATE	S. J. Kolot 21/1/77 D.M. TURNER 25-5-79 WOODS 25-5-79 J. B. B. 25-5-79	INSTRUMENTATION T.E.P. SOUTH AUSTRALIA	A3 T0344 SHEET 2-7-1
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SPECTRUM ANALYSER

TELE TYPE INTERFACE



DEVICE No	TYPE	+5V PIN CON	DIG. GND PIN CON
IC1	8251	26	4
IC2	DM7406	14	7

PIN No	CONNECTIONS TO 'D' TYPE CONNECTOR
1	+5V
2	DIG GND
3	TTY Rx +
4	TTY Rx RETURN -
5	NOT USED
6	TTY Tx RETURN -
7	TTY Tx
8	NOT USED
9	NOT USED

DRN  
CHKD  
INS'D  
AUTH

D.M. TURNER  
M. J. B. L. K.  
M. J. B. L. K.  
M. J. B. L. K.

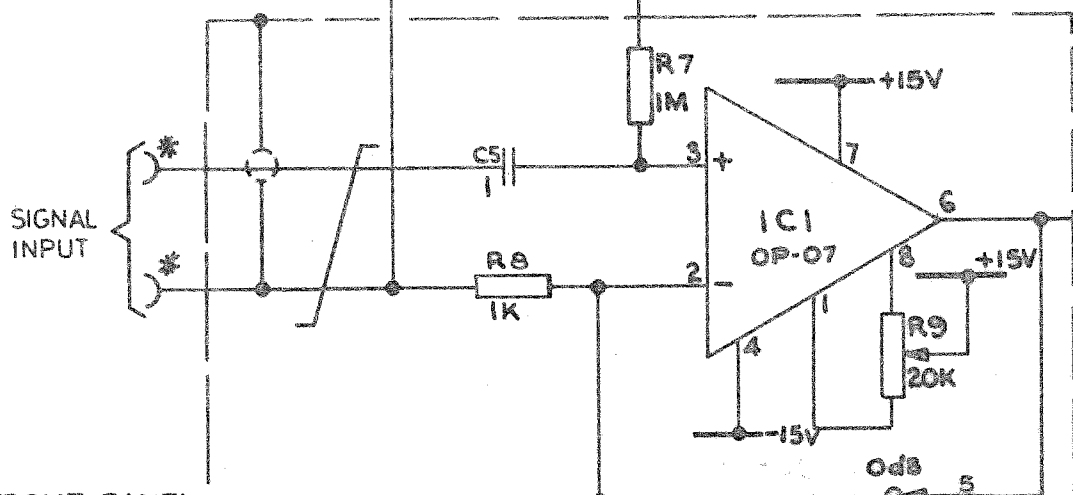
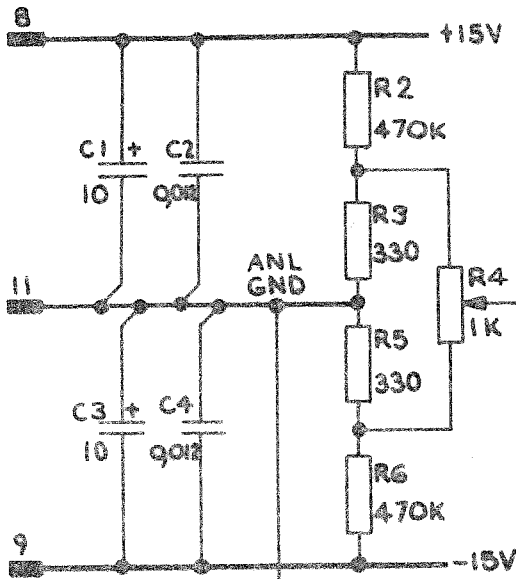
THE ELECTRICITY TRUST OF SOUTH AUSTRALIA

SCALE  
OP3

INSTRUMENTATION T.E.P.

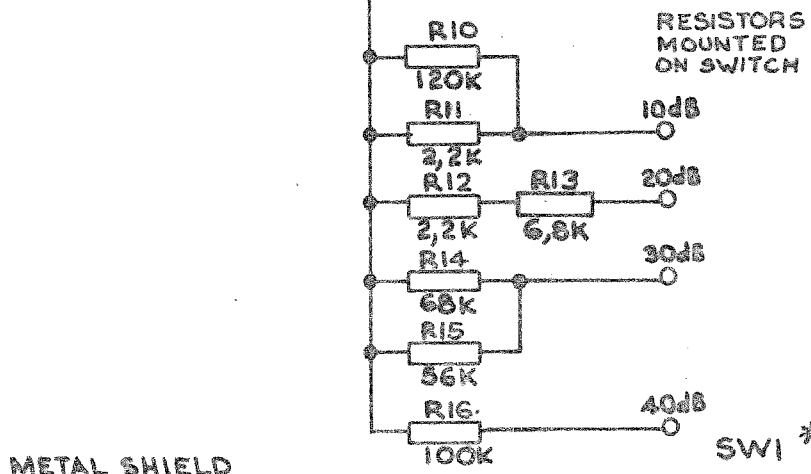
A4-T0344  
SHEET 2.8.1

I/P OVERLOAD INDICATOR



I/P1  
I/P2  
I/P3  
I/P4  
I/P5  
I/P6  
TO LOW PASS FILTER

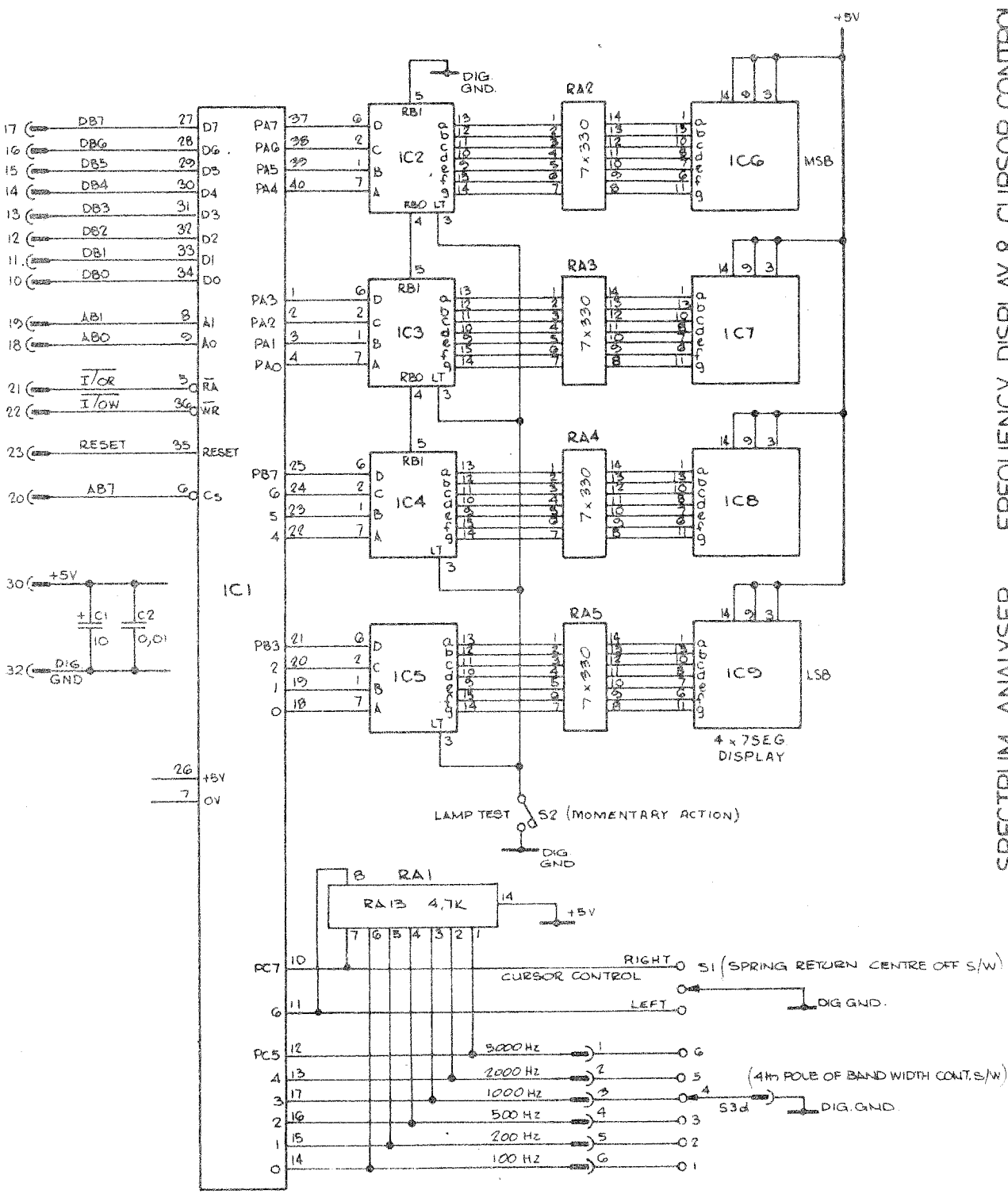
\* FRONT PANEL MOUNTED



METAL SHIELD

SPECTRUM ANALYSER INPUT CONDITIONING BOARD

DRN	D.M. TURNER 24-47	THE ELECTRICITY TRUST OF SOUTH AUSTRALIA	SCALE	OP3
CKD	<i>S.D. Robert</i> 5/1/77	INSTRUMENTATION T.E.P.	A4-T0344 SHEET 2.9.1	
INSP	<i>Wendy</i> 25.5.79			
AUTH	<i>Wendy</i> 27.5.79			

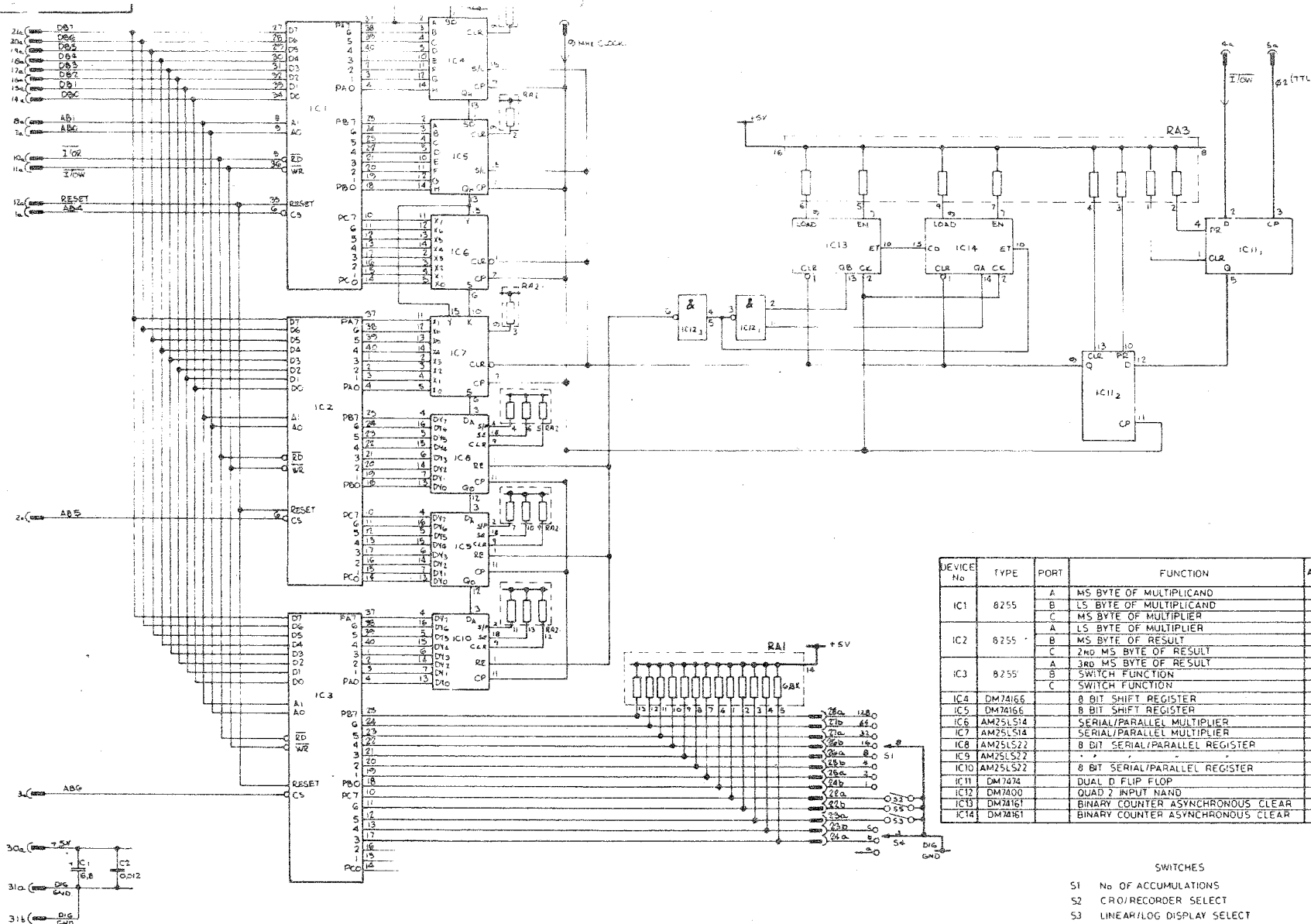


DEVICE No	TYPE	PORT	ADDRESS	MODE	+5V	DIG GND
IC1	8255	A	#07CH		26	7
		B	#07DH			
		C	#07EH			
			#07FH			
IC2-5	7446				16	8
IC6-9	DL707				3,9,14	-

SPECTRUM ANALYSER FREQUENCY DISPLAY & CURSOR CONTROL BOARD

THE ELECTRICITY TRUST OF SOUTH AUSTRALIA	SCALE
1:5.79	2:5.79
S. K. WARDLAW	7/16/79
S. J. GIBSON	2/15/79
S. J. GIBSON	2/15/79
INSTRUMENTATION T.F.P.	
A3 T0344	





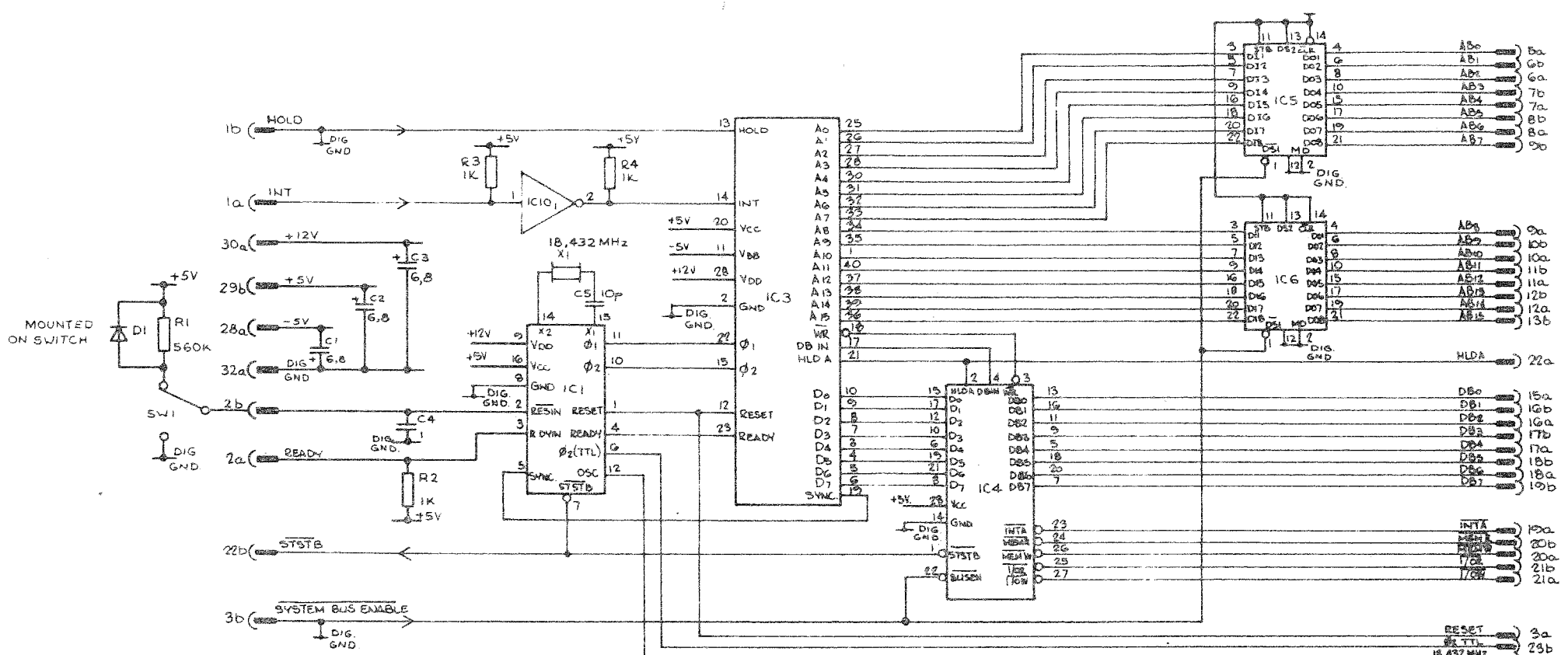
DEVICE No	TYPE	PORT	FUNCTION	ADDRESS	MODE	DIG GND	+5V	1K
IC1	8255	A	MS BYTE OF MULTIPLICAND	#EC	#EF	7	26	
		B	LS BYTE OF MULTIPLICAND	#ED				
		C	MS BYTE OF MULTIPLIER	#EE				
IC2	8255	A	LS BYTE OF MULTIPLIER	#DC	#DF	7	26	
		B	MS BYTE OF RESULT	#DD				
		C	2ND MS BYTE OF RESULT	#DE				
IC3	8255	A	3RD MS BYTE OF RESULT	#BC	#BF	7	26	
		B	SWITCH FUNCTION	#BD				
		C	SWITCH FUNCTION	#BE				
IC4	DM74166		8 BIT SHIFT REGISTER			8,6	16	9
IC5	DM74166		8 BIT SHIFT REGISTER			8,6	16	9
IC6	AM25LS14		SERIAL/PARALLEL MULTIPLIER			8,10,9	16	
IC7	AM25LS14		SERIAL/PARALLEL MULTIPLIER			8	16	9
IC8	AM25LS22		8 BIT SERIAL/PARALLEL REGISTER			10,17,19,8	20	218,9
IC9	AM25LS22		8 BIT SERIAL/PARALLEL REGISTER			10,17,19,8	20	218,9
IC10	AM25LS22		8 BIT SERIAL/PARALLEL REGISTER			10,17,19,8	20	218,9
IC11	DM7474		DUAL D FLIP FLOP			7	14	1,4,10,13
IC12	DM7400		QUAD 2 INPUT NAND			7	14	
IC13	DM74161		BINARY COUNTER ASYNCHRONOUS CLEAR			8	15	7,9
IC14	DM74161		BINARY COUNTER ASYNCHRONOUS CLEAR			8	16	7,9

- SWITCHES
- S1 No OF ACCUMULATIONS
  - S2 CRO/RECORDER SELECT
  - S3 LINEAR/LOG DISPLAY SELECT
  - S4 WINDOW SELECT
    - a) NO WINDOW
    - b) BINGHAM WINDOW
    - c) HANNING WINDOW
  - S5 SPARE (TO BE MAGNETUDE/FREQ. SELECT)

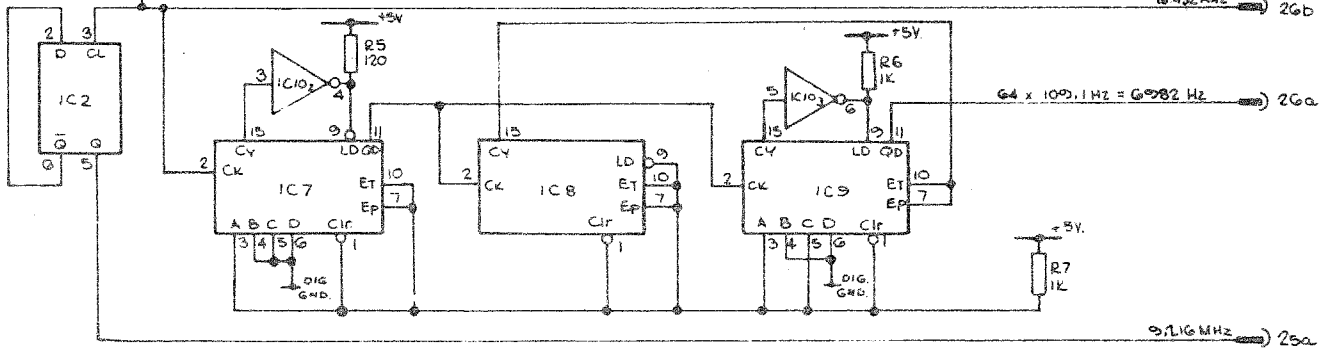
SPECTRUM ANALYSER HARDWARE MULTIPLIER

THE ELECTRICITY TRUST OF SOUTH AUSTRALIA				SCALE	OP3
INSTRUMENTATION T.E.P.				A2 - T0344	
				SHEET 2-11-1	

REV	DETAILS OF REVISION	RVD	CKD	APD	DATE	AUTH	DATE
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DEVICE No	TYPE	FUNCTION	+12V	+5V	-5V	DIG GND
1	8224	CLOCK GENERATOR	(C5)	(C4)	(C3)	8
2	7474	DUAL D FLIP FLOP		14,4,1		7
3	8080A	CPU	28	20	11	2
4	8228	SYSTEM CONTROLLER		28		14
5,6	8212	8 BIT I/O PORT		24		12,2
7	93516	HIGH SPEED BINARY COUNT.		16		8,4,5,6
8	74161-934	BINARY COUNTER		16		8
9	74161-934	BINARY COUNTER		16		8,4,6
10	7406	HEX. INVERTER		14		7



SPECTRUM ANALYSER CPU BOARD

SK Warburton 2579  
 J.P. Roberts 71578  
 Woodhall 25579  
 Leland 21579

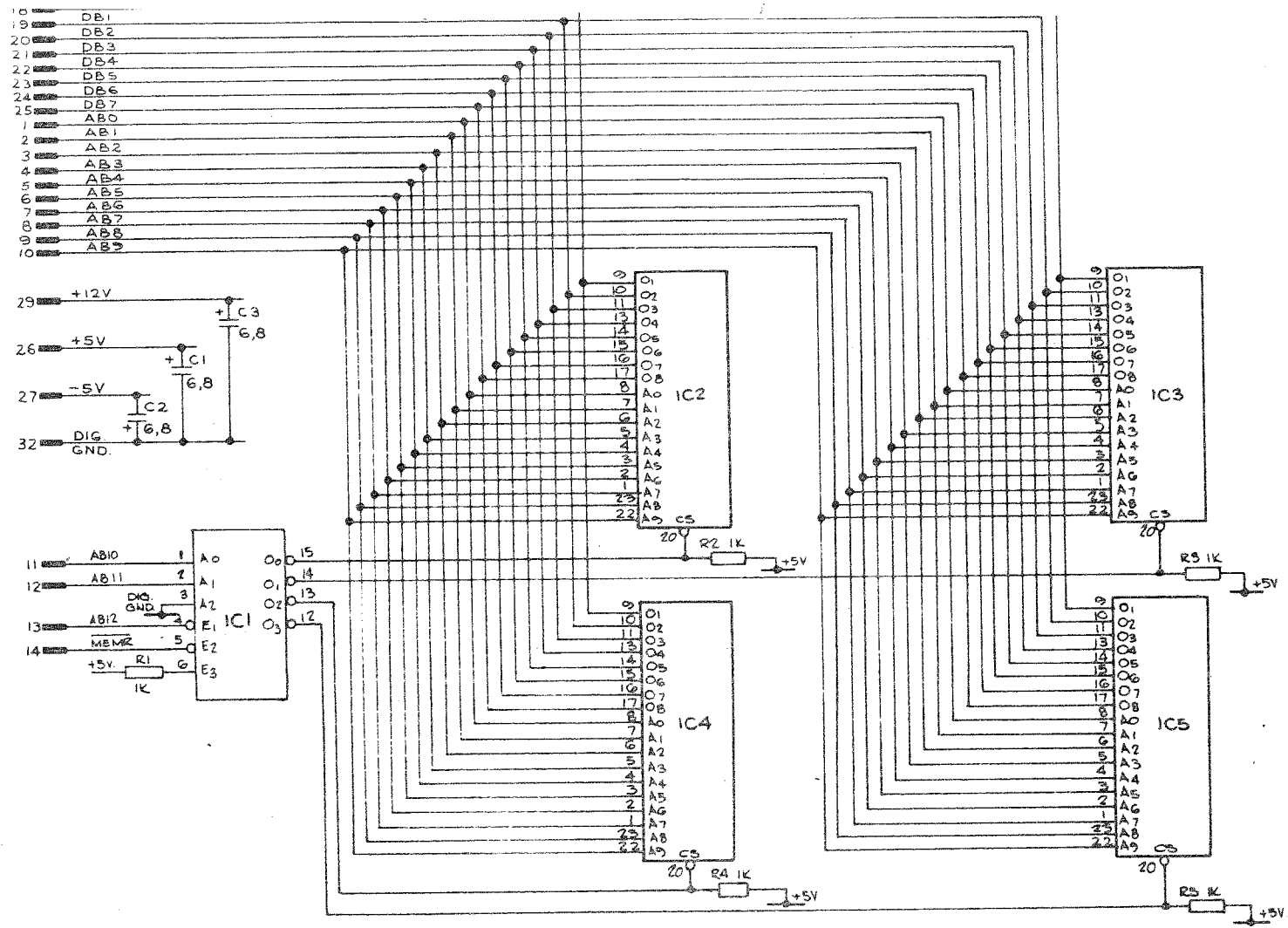
THE INSTRUMENTATION T.E.P. GROUP

INSTRUMENTATION T.E.P.

A3-T0344 SHEET 2-12-1

60 OF 3





DEVICE No	TYPE	FUNCTION	MEMORY	ADDRESS	+12V	+5V	-5V	DIG GND
IC1	8205	BINARY 1 OF 8 DECODER			(C3)	(C1)	(C2)	8,3
IC2	8708	1K x 8 BIT EPROM	0000	03FF	19	24	21	12,18
IC3	8708	1K x 8 BIT EPROM	0400	07FF	19	24	21	12,18
IC4	8708	1K x 8 BIT EPROM	0800	0BFF	19	24	21	12,18
IC5	8708	1K x 8 BIT EPROM	0C00	0FFF	19	24	21	12,18

SPECTRUM ANALYSER

R.O.M. BOARD

S.K. Warburton 1.5.79 J.P. Holst 3/8/79 W. ... 25.8.79 J. ... 12.8.79		INSTRUMENTATION T.E.P.	SHEET 2. 14. 1
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APPENDIX 2LISTING OF ASSEMBLY AND MACHINE CODE

0200		RD	SET	0200H	
02F5		LIN	SET	02F5H	
0400		MC	SET	0400H	
0434		LDG	SET	0434H	
051A		OUTU	SET	051AH	
0598		REC	SET	0598H	
0000			ORG	0000H	
0000	3E8A	BEGIN:	MVI	A,8AH	
0002	D3F7		OUT	0F7H	
0004	3F80		MVI	A,80H	
0006	D3EF		OUT	0EFH	
0008	3E88		MVI	A,88H	
000A	D30F		OUT	0DFH	
000C	3F89		MVI	A,89H	
000E	D37F		OUT	7FH	
0010	3E98		MVI	A,98H	
0012	D38F		OUT	08FH	*SET MODE OF PIA CHIPS
0014	318F13		LXI	SP,138FH	*SET STACK POINTER
0017	210010		LXI	H,1000H	
001A	0600		MVI	B,0	
001C	3F13		MVI	A,13H	
001F	70	ZERO:	MOV	M,B	
001F	23		INX	H	
0020	8C		CMP	H	
0021	C21E00		JNZ	ZERO	:(1000-12FF)=0
0024	0980		IN	080H	*READ ACCUMULATION SELECTOR SWITCH
0026	2F		CMA		
0027	32FF13		STA	13FEH	*VARIABLE
002A	0602		MVI	B,2	
002C	04	SOT:	INR	B	
002D	17		RAL		
002E	D22C00		JNC	SOT	
0031	78		MOV	A,B	
0032	32FF13		STA	13FFH	:(13FF)=2+LOG2(256/NO.ACC)
0035	088E		IN	08EH	
0037	F618		ANI	18H	
0039	32FD13		STA	13FDH	*WINDOW SELECTION

003C	087E		IN	7EH	: READ FREQUENCY SELECTOR SWITCH
003F	2F		CMA		: AND CALCULATE MULTIPLICATION FACTOR
003F	E63F		ANI	3FH	: SELECT 6 LS BITS
0041	0501		MVI	R+1	
0043	1F		RAR		
0044	DA6100		JC	MF	
0047	0602		MVI	R+2	
0049	1F		RAR		
004A	DA6100		JC	MF	
004D	0605		MVI	R+5	
004E	1F		RAR		
0050	DA6100		JC	MF	
0053	060A		MVI	R+10	
0055	1F		RAR		
0056	DA6100		JC	MF	
0059	0614		MVI	R+20	
005A	1F		RAR		
005C	DA6100		JC	MF	
005F	0632		MVI	R+50	
0061	78	MF:	MOV	A+8	
0062	32FA13		STA	13F8H	
0065	C0DF00	ACCUH:	CALL	START	
0068	00		NOP		
0069	3AFD13		LDA	13FDH	
006C	FE08		CPI	8H	
006F	CC3001		CZ	HAMND	: IS HANNING WINDOW SELECTED?
0071	3AFD13		LDA	13FDH	
0074	FE10		CPI	10H	
0076	CC6201		CZ	HAMMO	: IS PARTIAL WINDOWING SELECTED?
0079	00		NOP		
007A	C08801		CALL	SHUFF	: SHUFFLE DATA
007D	C00002		CALL	R0	: CALCULATE FOURIER COEFFICIENTS
0080	00		NOP		
0081	C00004		CALL	WC	: CORRECT AMPLITUDE AND ACCUMULATE
0084	00		NOP		
0085	D8BE		IN	08EH	
0087	17		RAL		

0085	D2A600		JNC	SKIP	*CR0 SELECTED?
0088	DBRE		IN	DBEH	
008D	E620		ANI	20H	
008F	CCR404		CZ	LOG	*LOG SELECTED?
0092	DBRE		IN	DBEH	
0094	F620		ANI	20H	
0096	C4F502		CNZ	LIN	*LINEAR SELECTED?
0099	CD1A05		CALL	OUT0	*DISPLAY ON CR0 FOR .5S
009C	CD1A05		CALL	OUT0	
009F	CD1A05		CALL	OUT0	
00A2	CD1A05		CALL	OUT0	
00A5	00		MOR		
00A6	3AFE13	SKIP:	LDA	13FEH	
00A9	30		OCR	A	
00AA	32FE13		STA	13FEH	*A=A-1
00AD	C26500		JNZ	ACCUM	*IF(A.NE.0)GOTO ACCUM
00B0	DBRE	REAP:	IN	DBEH	
00B2	E620		ANI	20H	
00B4	32FC13		STA	13FCH	*LOG-LIN IN 13FCH
00B7	CCR404		CZ	LOG	*LOG SELECTED?
00BA	DBRE		IN	DBEH	
00BC	E620		ANI	20H	
00BE	C4F502		CNZ	LIN	*LINEAR SELECTED?
00C1	DBRE		IN	DBEH	
00C3	17		RAL		
00C4	D29805		JMC	REC	*RECORDER SELECTED?
00C7	CD1A05	CROR:	CALL	OUT0	*DISPLAY ON CR0 FOR 0.125 SECS.
00CA	DBRE		IN	DBEH	
00CC	17		RAL		
00CD	D2B000		JMC	REAP	*RECORDER SELECTED?
00D0	DBRE		IN	DBEH	
00D2	F620		ANI	20H	
00D4	47		MOV	B+A	
00D5	3AFC13		LDA	13FCH	
00D8	88		CMP	B	
00D9	CAC700		JZ	CROR	*LOG-LINEAR UNCHANGED?
00DC	C3E000		JMP	REAP	



000F	210014	START:	LXI	H,1400H	: A/D CONVERSION
00E2	08F6	A0:	IN	0F6H	: SYNCHRONISE BEFORE STARTING
00E4	E620		ANI	20H	
00F6	C2E200		JNZ	A0	
00E9	08F6	A1:	IN	0F6H	
00ER	E620		ANI	20H	
00ED	CAE900		JZ	A1	:PC5=0?
00F0	00		NOP		
00F1	00		NOP		
00F2	08F5		IN	0F5H	
00F4	77		MOV	M,A	: (M)=PORT A
00F5	24		INX	H	:H=15H
00F6	08F6		IN	0F6H	
00F8	F6C0		ANI	0C0H	
00FA	77		MOV	M,A	:MS 2 BITS SELECTED
00FB	25		DCR	H	:H=14H
00FC	2C		INP	L	:L=L+1
00FD	CA0A01		JZ	A3	
0100	08F6	A2:	IN	0F6H	
0102	E620		ANI	20H	
0104	CAE900		JZ	A1	:PC5=0?
0107	C30001		JMP	A2	
010A	210016	A3:	LXI	H,1600H	:H=16H,L=0
010D	0600		MVI	B,0	
010F	70	A4:	MOV	M,B	
0110	23		INX	H	
0111	3E18		MVI	A,18H	
0113	BC		CMP	H	
0114	C20F01		JNZ	A4	:H,ME,18H?
0117	2614		MVI	H,14H	:CHECK INPUT DATA FOR OVERLOAD
0119	067F		MVI	H,7FH	
011B	0E80		MVI	C,80H	
011D	7F	A5:	MOV	A,M	
011E	8B		CMP	R	
011F	CA2B01		JZ	A6	
0122	99		CMP	C	
0123	CA2B01		JZ	A6	

0126	2C		INR	L	
0127	C21001		JNZ	45	
012A	C9		RET		
012B	3E08	46:	MVI	A,8	
012D	D3F6		OUT	0F6H	
012F	76		HLT		
0130	16C0	HANND:	MVI	0+0C0H	;D=0C0H
0132	1E00		MVI	F,0	;E=0
0134	2606	HANNI:	MVI	H,6	;H=6
0136	6A		MOV	L,D	
0137	7E		MOV	A,M	
0138	D3EC		OUT	0ECH	
013A	24		INR	H	;H=7
013B	7E		MOV	A,M	
013C	D3E0		OUT	0EDH	;MULTIPLICAND= (1/2) SIN(D)
013E	2614		MVI	H,14H	;H=14H
0140	68		MOV	L,E	
0141	7E		MOV	A,M	
0142	47		MOV	B,A	
0143	D3FE		OUT	0EEH	
0145	24		INR	H	;H=15H
0146	7E		MOV	A,M	
0147	4F		MOV	C,A	
0148	D30C		OUT	0DCH	;MULTIPLIER=XRE
014A	78		MOV	A,R	
014B	FF80		CFI	80H	
014D	3F		CMC		
014E	1F		RAR		
014F	47		MOV	B,A	
0150	79		MOV	A,C	
0151	1F		RAR		
0152	4F		MOV	C,A	; (RC) = XRE/2
0153	DBDE		IN	0DEH	;2ND MS RESULT
0155	81		ADD	C	
0156	71		MOV	M,C	
0157	25		DCR	H	;H=14H
0158	DB0D		IN	0DDH	;MS RESULT

0154	88		ADC	B	
0158	77		MOV	M, A	: XRE = (XRE + SIN(D)) / 2
015C	14		INC	D	: D = D + 1
015D	1C		INC	E	: E = E + 1
015E	C23401		JNZ	HANN1	: E = NE * 0?
0161	C9		RFT		
0162	16C0	HANN0:	MVI	D, 0C0H	: D = 0C0H
0164	1F00		MVI	E, 0	: E = 0
0166	2606	HANN1:	MVI	H, 6	: H = 6
0168	6A		MOV	L, 0	
0169	7F		MOV	A, M	
016A	D3EC		OUT	0ECH	
016C	24		INC	H	: H = 7
016D	7E		MOV	A, M	
016E	D3ED		OUT	0EDH	: MULTIPLICAND = (1/2) SIN(D)
0170	2614		MVI	H, 14H	
0172	6B		MOV	L, E	
0173	7F		MOV	A, M	
0174	47		MOV	B, A	
0175	D3EE		OUT	0EEH	
0177	24		INC	H	: H = 15H
0178	7E		MOV	A, M	
0179	4F		MOV	C, A	
017A	D3DC		OUT	0DCH	: MULTIPLIER = XRE
017C	78		MOV	A, B	
017D	FAA0		CPI	60H	
017F	3F		CNC		
0180	1F		RAR		
0181	47		MOV	B, A	
0182	79		MOV	A, C	
0183	1F		RAR		
0184	4F		MOV	C, A	: (BC) = XRE / 2
0185	DADE		IN	0DEH	: 2ND MS RESULT
0187	81		ADD	C	
0188	77		MOV	M, A	
0189	25		DCR	H	: H = 14H
018A	DAFD		IN	0DFH	: MS RESULT

018C	88	ADC	B	
018D	77	MOV	H,A	:(HE) = ((HE) + SIN(D)) / 2
018E	97	SUB	A	
018F	93	SUB	E	
0190	6F	MOV	L,A	:L = -E
0191	7E	MOV	A,M	
0192	47	MOV	R,A	
0193	03EE	OUT	0EEH	:MS MULTIPLIER
0195	24	INR	H	:H = 15H
0196	7E	MOV	A,M	
0197	4F	MOV	C,A	
0198	030C	OUT	00CH	:LS MULTIPLIER
019A	78	MOV	A,B	
019B	FE80	CPI	80H	
019D	3F	CMC		
019E	1F	RAR		
019F	47	MOV	B,A	
01A0	79	MOV	A,C	
01A1	1F	RAR		
01A2	4F	MOV	C,A	:(BC) ARE HALVED
01A3	0BDF	IN	0DEH	:2ND MS RESULT
01A5	81	ADD	C	
01A6	77	MOV	H,A	
01A7	25	DCR	H	:H = 14H
01A8	0BDD	IN	0DDH	:MS RESULT
01AA	88	ADC	B	
01AF	77	MOV	M,A	
01AC	74	MOV	A,D	
01AD	C604	ADJ	04H	
01AF	57	MOV	D,A	:D = D + 4
0180	1C	INR	E	
0181	76	MOV	A,F	
0182	FE20	CPI	20H	
0184	C26601	JNZ	H&MMI	:E,NE,20H?
0187	C9	RFT		
0188	210114	SHUF0:	LXI	H,1401H :H=14H,L=1
0188	55	SHUF1:	MOV	D,L :D=L

```

018C 1E0A      MVI    E,8H      ;E=8
018F 7A      SHUF2:  MOV    A,D
019F 07      RLC
01C0 57      MOV    D,A      ;ROLL D LEFT
01C1 78      MOV    A,R
01C2 1F      RAR
01C3 47      MOV    B,A      ;ROLL R RIGHT THRU CARRY
01C4 1D      DCR    F      ;F=F-1
01C5 C2#F01    JNZ    SHUF2     ;E.NE.0?
01C8 7D      MOV    A,L
01C9 88      CMP    B
01CA D2#D01    JNC    SHUF3     ;L.GTE.R?
01CD 56      MOV    D,M
01CE 4D      MOV    C,L
01CF 68      MOV    L,R
01D0 5E      MOV    E,M
01D1 72      MOV    M,D
01D2 69      MOV    L,C
01D3 73      MOV    M,E      ;(HL) AND (HR) ARE SWAPPED
01D4 24      INR    H      ;H=15H
01D5 56      MOV    D,M
01D6 4D      MOV    C,L
01D7 68      MOV    L,B
01D8 5E      MOV    E,M
01D9 72      MOV    M,D
01DA 69      MOV    L,C
01DB 73      MOV    M,E      ;(HL) AND (HR) ARE SWAPPED
01DC 25      DCR    H      ;H=14H
01DD 2C      SHUF3:  INR    L      ;L=L+1
01DE C2#B01    JNZ    SHUF1     ;L.NE.0?
01E1 C9      RET

```

NO PROGRAM ERRORS

\* 01

A	0007	A0	00E2	A1	00E9	A2	0100
A3	010A	A4	010F	A5	0110	A6	012B
ACCUM	0065	B	0000	B0	0200	BEGIN	0000 *
C	0001	C0R	00C7	D	0002	E	0003

H	0004	HANMO	0162	HANMI	0166	HANMO	0130
HANMI	0134	L	0005	LIN	02F5	LOB	04B4
K	0006	MF	0061	OUT0	051A	PSW	0006
REAP	0080	REC	0598	SHUF0	01B8	SHUF1	01B8
SHUF2	01FF	SHUF3	0100	SKIP	00A6	SOT	002C
SP	0006	START	00DF	WC	0400	ZERO	001E

0200			ORG	200H	
0200	010180	80:	LXI	B,8001H	:B=80H,C=1
0203	2E00	91:	MVI	L,0	:L=0
0205	55	82:	MOV	D,L	:D=L
0206	1E00		MVI	E,0	:E=0
0208	E5	83:	PUSH	H	:SAVE L IN STACK
0209	2605		MVI	H,6	
020B	6B		MOV	L,E	
020C	7E		MOV	A,M	
020D	03EC		OUT	03EH	
020F	24		INR	H	:H=7
0210	7E		MOV	A,M	
0211	03ED		OUT	03EH	:SIN(E) MULTIPLICAND PORT
0213	7D		MOV	A,L	
0214	0640		ADI	40H	
0216	6F		MOV	L,A	
0217	7E		MOV	A,M	
0218	32C113		STA	13C1H	:LS BYTE OF COS(E) AT 13C1H
0218	25		DCR	H	:H=6
021C	7E		MOV	A,M	
021D	32C013		STA	13C0H	:MS BYTE OF COS(E) AT 13C0H
0220	E1		POP	H	:RECOVER L FROM STACK
0221	2614		MVI	H,14H	:H=14H
0223	7D		MOV	A,L	
0224	81		ADD	C	
0225	6F		MOV	L,A	:L=L+C
0226	7E		MOV	A,M	
0227	32C213		STA	13C2H	:MS BYTE OF A(L+C) AT 13C2H
022A	03EE		OUT	03EH	
022C	24		INR	H	
022D	7E		MOV	A,M	
022E	32C313		STA	13C3H	:LS BYTE OF A(L+C) AT 13C3H
0231	03DC		OUT	03EH	:AL+C IN MULTIPLIER PORT
0233	24		INR	H	:H=16H
0234	7E		MOV	A,M	
0235	32C413		STA	13C4H	:MS BYTE OF AL+C AT 13C4H
0238	24		INR	H	:H=17H

0239	7E	MOV	A,M	
023A	32C513	STA	13C5H	;LS BYTE OF HL+C AT 13C5H
023D	7D	MOV	A,L	
023E	91	SUB	C	
023F	6F	MOV	L,A	;L IS RESTORED
0240	2614	MVI	H,14H	
0242	7E	MOV	A,M	
0243	FE80	CPI	80H	
0245	3F	CMC		
0246	1F	RAR		
0247	32CA13	STA	13CAH	;MS BYTE OF AL/2 AT 13CAH
024A	24	INR	H	;H=15H
024B	7E	MOV	A,M	
024C	1F	RAR		
024D	32CB13	STA	13CBH	;LS BYTE OF AL/2 AT 13CBH
0250	24	INR	H	;H=16H
0251	7E	MOV	A,M	
0252	FE80	CPI	80H	
0254	3F	CMC		
0255	1F	RAR		
0256	32CC13	STA	13CCH	;MS BYTE OF BL/2 AT 13CCH
0259	24	INR	H	;H=17H
025A	7E	MOV	A,M	
025B	1F	RAR		
025C	32CD13	STA	13CDH	;LS BYTE OF BL/2 AT 13CDH
025F	79	MOV	A,C	
0260	32CE13	STA	13CEH	;C STORED AT 13CEH
0263	C5	PUSH	B	
0264	D5	PUSH	D	;SAVE VARIABLES
0265	DBDE	IN	0DEH	
0267	4F	MOV	C,A	
0268	DBDD	IN	0DDH	
026A	47	MOV	B,A	;BC=0.5AL+CSIN(E)
026B	3AC413	LDA	13C4H	
026E	D3EE	OUT	0EEH	
0270	3AC513	LDA	13C5H	
0273	03DC	OUT	0DCH	;BL+C IN MULTIPLIER



0275	ORDE	IN	00EH	
0277	5F	MOV	E,A	
0278	DBDD	IN	00DH	
027A	57	MOV	D,A	;DE=0.5BL+CSIN(E)
027B	3AC013	LDA	13C0H	
027E	03EC	OUT	0ECH	
0280	3AC113	LDA	13C1H	
0283	03ED	OUT	0EDH	
0285	DRDE	IN	0DEH	
0287	91	SUB	C	
0288	4F	MOV	C,A	
0289	DBDD	IN	0DDH	
028A	98	SAR	H	
028C	47	MOV	R,A	;RC=(BL+CCOS(E)-AL+CSIN(E))/2
028D	3AC213	LDA	13C2H	
0290	03EE	OUT	0EEH	
0292	3AC313	LDA	13C3H	
0295	03DC	OUT	0DCH	
0297	0BDE	IN	0DEH	
0299	83	ADD	E	
029A	5F	MOV	E,A	
029B	DBDD	IN	0DDH	
029D	8A	ADC	D	
029F	57	MOV	D,A	;DE=(AL+CCOS(E)+BL+CSIN(E))/2
029F	2615	MVI	H,15H	
02A1	3ACR13	LDA	13CRH	
02A4	83	ADD	E	
02A5	77	MOV	M,A	
02A6	25	DCR	H	;H=14H
02A7	3ACA13	LDA	13CAH	
02AA	8A	ADC	D	
02AB	77	MOV	M,A	;AL=(AL+(AL+CCOS(E)+BL+CSIN(E)))/2
02AC	2617	MVI	H,17H	
02AE	3ACD13	LDA	13CDH	
02B1	81	ADD	C	
02B2	77	MOV	M,A	
02B3	25	DCR	H	;H=16H

0284	3ACC13	LDA	13CCH	
0287	88	ADC	H	
0288	77	MOV	M,A	:BL=(HL+(BL+CCOS(E)-AL+CSIN(E)))/2
0289	3ACE13	LDA	13CEH	:VARIABLE C IS RECOVERED
028C	85	ADD	L	
028D	6F	MOV	L,A	:L=L+C
028E	24	INR	H	:H=17H
028F	3ACD13	LDA	13CDH	
02C2	91	SUB	C	
02C3	77	MOV	M,A	
02C4	25	DCR	H	:H=16H
02C5	3ACC13	LDA	13CCH	
02C8	98	SRB	B	
02C9	77	MOV	M,A	:BL+C=(BL-(BL+CCOS(E)-AL+CSIN(E)))/2
02CA	25	DCR	H	:H=15H
02CB	3ACB13	LDA	13CBH	
02CE	93	SUB	E	
02CF	77	MOV	M,A	
02D0	25	DCR	H	:H=14H
02D1	3ACA13	LDA	13CAH	
02D4	9A	SRB	D	
02D5	77	MOV	M,A	:AL+C=(AL-(AL+CCOS(E)+BL+CSIN(E)))/2
02D6	D1	POP	D	
02D7	C1	POP	B	:RECOVER VARIABLES FROM STACK
02D8	7D	MOV	A,L	
02D9	91	SUB	C	
02DA	6F	MOV	L,A	
02DB	2C	INR	L	:L=L+1
02DC	78	MOV	A,B	
02DD	83	ADD	E	
02DE	5F	MOV	E,A	:E=E+B
02DF	79	MOV	A,C	
02E0	82	ADD	D	
02E1	8D	CMP	L	
02E2	C20802	JNZ	B3	:L.NE.(D+C)?
02E5	79	MOV	A,C	
02E6	85	ADD	L	

02E7	6F		MOV	L,A	:L=L+C
02E8	C20502		JNZ	B2	:L.NE.0?
02E9	78		MOV	A,B	
02EC	0F		RRC		
02ED	47		MOV	R,A	:R=R/2
02EE	79		MOV	A+C	
02EF	87		ADD	A	
02F0	4F		MOV	C,A	:C=2C
02F1	C20302		JNZ	H1	:C.NE.0?
02F4	C9		RET		
02F5	2E00	LIN:	MVI	L,0	:L=0
02F7	3EAA		MVI	A,0AAH	
02F9	03ED		OUT	0EDH	
02FB	3E6B		MVI	A,6BH	
02FD	03EC		OUT	0ECH	:MULT FACTOR IN MULTIPLICAND
02FF	E5	SORT:	PUSH	H	
0300	2610		MVI	H,10H	
0302	7E		MOV	A,M	
0303	03EE		OUT	0EEH	
0305	24		INR	H	:H=11H
0306	7E		MOV	A,M	
0307	03DC		OUT	0DCH	:PSD IN MULTIPLIER
0309	0B00		IN	0DDH	
030B	57		MOV	D,A	
030C	0BDE		IN	0DEH	
030E	5F		MOV	E,A	
030F	0BBC		IN	0BCH	
0311	47		MOV	B,A	:DER=RESULT
0312	0F02		MVI	C,2	:C=2
0314	78	PREM:	MOV	A,B	
0315	87		ADD	A	
0316	47		MOV	B,A	
0317	78		MOV	A,E	
0318	8F		ADC	A	
0319	5F		MOV	E,A	
031A	7A		MOV	A,0	
031B	8F		ADC	A	

031C	57		MOV	D,A	:DER=2*DFB
031D	0D		OCR	C	:C=C-1
031F	C21403		JNZ	PREMA	:C.NF.0?
0321	3E08		MVI	A,BH	
0323	32C013		STA	13C0H	:S=R
0326	97		SUB	A	
0327	32C113		STA	13C1H	:T=0
032A	67		MOV	H,A	
032B	6F		MOV	L,A	:HL=0
032C	010100		LXI	R,1	:U=RC=1
032F	3AC113	SORT1:	LDA	13C1H	
0332	87		ADD	A	
0333	32C113		STA	13C1H	:T=2T
0336	7E		MOV	A,E	
0337	87		ADD	A	
0338	5F		MOV	E,A	
0339	7A		MOV	A,D	
033A	8F		ADC	A	
033B	57		MOV	D,A	
033C	7D		MOV	A,L	
033D	8F		ADC	A	
033E	6F		MOV	L,A	
033F	7C		MOV	A,H	
0340	8F		ADC	A	
0341	67		MOV	H,A	
0342	7E		MOV	A,E	
0343	87		ADD	A	
0344	5F		MOV	E,A	
0345	7A		MOV	A,D	
0346	8F		ADC	A	
0347	57		MOV	D,A	
0348	7D		MOV	A,L	
0349	8F		ADC	A	
034A	6F		MOV	L,A	
034B	7C		MOV	A,H	
034C	8F		ADC	A	
034D	67		MOV	H,A	:R=49

034E	70	MOV	A,L	
034F	91	SUB	C	
0350	7C	MOV	A,H	
0351	96	SHR	R	
0352	FA6403	JM	SHORT2	:RC.BT.HL?OR U.GT.R?
0355	7D	MOV	A,L	
0356	91	SUB	C	
0357	6F	MOV	L,A	
0358	7C	MOV	A,H	
0359	98	SHR	R	
035A	67	MOV	H,A	:R=R-U
035B	3AC113	LDA	13C1H	
035E	3C	INR	A	
035F	32C113	STA	13C1H	:T=T+1
0362	03	INX	H	
0363	03	INX	R	:U=U+2
0364	79	MOV	A,C	SHORT2:
0365	87	ADD	A	
0366	4F	MOV	C,A	
0367	78	MOV	A,H	
0368	8F	ADC	A	
0369	47	MOV	R,A	
036A	0B	DCX	R	:U=2U-1
036B	3AC013	LDA	13C0H	
036E	3D	DCR	A	
036F	32C013	STA	13C0H	:S=S-1
0372	C22F03	JNZ	SHORT1	:S.ME.0?
0375	3AC113	LDA	13C1H	
0378	H1	POP	H	
0379	2613	MVI	H,13H	
037B	77	MOV	M,A	:STORE T AT 13L
037C	2C	INR	L	:L=L+1
037D	7D	MOV	A,L	
037E	FE80	CPI	80H	
0380	C2FF02	JNZ	SHORT	:L.ME.80H?
0383	C9	RET		

NO PROGRAM ERRORS

\* 01

A	0007	R	0000	R0	0200 *	H1	0203
B2	0205	R3	0208	C	0001	D	0002
E	0003	H	0004	L	0005	LIN	02F5 *
M	0006	PREM8	0314	PS#	0006	SP	0006
SOPT	02FF	SOPT1	032F	SOPT2	0364		

0400			ORG	400H	
0400	3AFD13	MC:	LDA	13FDH	: AMPLITUDE CORRECTION DUE TO WINDOWS
0403	FE13		CPI	10H	
0405	C45C04		JZ	ACRE0	: NO WINDOW SELECTED ?
0408	FE03		CPI	8	
040A	CA4604		JZ	WH	: HANNING WINDOW SELECTED ?
040C	2F00		MVI	L,0	: NO-BINGHAM WINDOW
040F	3E49		MVI	A,49H	: MULTIPLICAND=2/7*256**2
0411	D3FC		OUT	0E0H	
0413	3E25		MVI	A,25H	
0415	D3E0		OUT	0E0H	
0417	2618	WB:	MVI	H,18H	
0419	25	WB1:	DCR	H	
041A	7E		MOV	A,H	
041B	D30C		OUT	0D0H	
041D	25		DCR	H	
041E	7E		MOV	A,H	
041F	D3EE		OUT	0EEH	
0421	D60D		IN	0D0H	
0423	47		MOV	B,A	
0424	D8DE		IN	0DEH	
0426	4F		MOV	C,A	
0427	D8FC		IN	0F0H	
0429	57		MOV	D,A	: BC=2/7*X(L)
042A	1F02		MVI	E,2	
042C	7A	WB2:	MOV	A,D	
042D	87		ADD	A	
042F	57		MOV	D,A	
042F	79		MOV	A,C	
0430	8F		ADD	A	
0431	4E		MOV	C,A	
0432	78		MOV	A,B	
0433	88		ADD	B	
0434	47		MOV	B,A	: RCD=2*B*CD
0435	1D		DCR	E	
0436	C22C04		JNZ	WB2	
0439	7C		MOV	A,H	

043A	FF14		CPI	14H	
043C	021904		JNZ	WH1	; H=14H?
043E	20		DCR	L	
0440	021704		JNZ	WS	; L.NE0?
0443	035C04		JMP	ACRE0	
0446	2F00	WH1:	MVI	L,0	
0448	2615	WH1:	MVI	H,18H	
044A	25	WH2:	DCR	H	
044C	7E		MOV	A,M	
044D	87		ADD	A	
044E	77		MOV	M,A	
044F	25		DCR	H	
0450	7E		MOV	A,M	
0451	8F		ADC	A	
0452	77		MOV	M,A	; X(L)=2*X(L)
0453	7C		MOV	A,H	
0454	FF14		CPI	14H	
0455	024404		JNZ	WH2	
0456	20		DCR	L	
0457	024804		JNZ	WH1	
0458	2F00	ACRE0:	MVI	L,0	
0459	2614	ACRE1:	MVI	H,14H	
045A	7E		MOV	A,M	
045B	03EC		OUT	03CH	
045C	03EE		OUT	03EK	
045D	24		INR	H	; H=15H
045E	7E		MOV	A,M	
045F	03ED		OUT	03DH	
0460	03DC		OUT	03CH	
0461	0F00		IN	0DDH	
0462	4F		MOV	C,A	
0463	0BDE		IN	0DEH	
0464	57		MOV	D,A	
0465	0BFC		IN	0BCH	
0466	5F		MOV	E,A	; CDE=AL*AL
0467	24		INR	H	; H=16H
0468	7E		MOV	A,M	



0476	03EC		OUT	0E0H	
0478	03EE		OUT	0EEH	
047A	24		IMR	H	:H=17H
047B	7E		MOV	A,M	
047C	03E0		OUT	0E0H	
047E	03DC		OUT	0DC0	
0480	0HBC		IN	0BCH	
0482	83		ADD	E	
0483	5F		MOV	E,A	
0484	0HDE		IN	0DEH	
0486	8A		ADC	D	
0487	57		MOV	D,A	
0488	0B00		IN	0B0H	
048A	89		ADC	C	
048F	4F		MOV	C,A	:CDE=AL*AL+HL*BL
048C	3AFF13		LDA	13FFH	
048F	67		MOV	H,A	:H=(13FFH)
0490	0600		MVI	R,0	
0492	7C	ACR2:	MOV	A,E	
0493	87		ADD	A	
0494	5F		MOV	E,A	
0495	7A		MOV	A,0	
0496	8F		ADC	A	
0497	57		MOV	D,A	
0498	79		MOV	A,C	
0499	8F		ADC	A	
049A	4F		MOV	C,A	
049B	7B		MOV	A,B	
049C	8F		ADC	A	
049D	47		MOV	B,A	:BCDE=2*BCDE
049E	25		DCR	H	
049F	029204		JNZ	ACR2	:H.HE=0?
04A2	2612		MVI	H,12H	
04A4	7A		MOV	A,D	
04A5	86		ADD	M	
04A6	77		MOV	H,A	
04A7	2F		DCR	H	:H=11H

04A8	79		MOV	A,C	
04A9	8E		ADC	M	
04AA	77		MOV	H,A	
04AB	25		DCR	H	!H=10H
04AC	78		MOV	A,B	
04AD	8E		ADC	M	
04AF	77		MOV	M,A	!R=R+RC0
04AF	2C		INR	L	
04B0	C25E04		JNZ	ACRE1	!L.NE.0?
04B3	C9		RET		
04B4	2E00	LOG:	MVI	L,0	!L=0
04B6	2610	LOG0:	MVI	H,10H	!H=10H
04B8	46		MOV	B,M	
04B9	24		INR	H	!H=11H
04BA	4F		MOV	C,B	
04BB	24		INR	H	!H=12H
04BC	56		MOV	D,M	!BCD=ACC RESULT
04BD	1E02		MVI	E,210	!E=210
04BF	7A	LOG1:	MOV	A,0	
04C0	87		ADD	A	
04C1	57		MOV	D,A	
04C2	79		MOV	A,C	
04C3	8F		ADC	A	
04C4	4F		MOV	C,A	
04C5	76		MOV	A,R	
04C6	8F		ADC	A	
04C7	47		MOV	B,A	!SHIFT BCD LEFT
04C8	0A0904		JC	LOG2	!CARRY SET?
04CA	7B		MOV	A,E	
04CC	0F04		SRI	0AH	
04CF	5F		MOV	E,A	!E=E-10
04CF	FEF6		CPI	0F6H	
04D1	C2BF04		JNZ	LOG1	!E.NE.-10?
04D4	1F00		MVI	E,0	!E=0
04D6	C30F05		JMP	LOG3	
04D9	FF12	LOG2:	CPI	12H	
04DB	D40F05		JC	LOG3	!B.LTE.12H?

040E	1C		INR	E	
040F	FE26		CPI	26H	
04E1	0A0F05		JC	LOG3	:B.LTE.26H?
04E4	1C		INR	F	
04E5	FE38		CPI	38H	
04E7	0A0F05		JC	LOG3	:B.LTE.38H?
04E8	1C		INR	E	
04E9	FE52		CPI	52H	
04ED	0A0F05		JC	LOG3	:B.LTE.52H?
04F0	1C		INR	F	
04F1	FE6A		CPI	6AH	
04F3	0A0F05		JC	LOG3	:B.LTE.6AH?
04F6	1C		INR	E	
04F7	FE84		CPI	84H	
04F9	0A0F05		JC	LOG3	:B.LTE.84H?
04FC	1C		INR	E	
04FD	FEA0		CPI	0A0H	
04FF	0A0F05		JC	LOG3	:B.LTE.0A0H?
0502	1C		INR	E	
0503	FEFE		CPI	0FEH	
0505	0A0F05		JC	LOG3	:B.LTE.0FEH?
0508	1C		INR	E	
0509	FEDE		CPI	0DEH	
050B	0A0F05		JC	LOG3	:B.LTE.0DEH?
050E	1C		INR	E	
050F	2613	LOG3:	MVI	H.13H	:H=13H
0511	73		MOV	M.E	
0512	2C		INR	L	
0513	3E80		MVI	A.80H	
0515	80		CMP	L	
0516	C28504		JNZ	LOG0	
0519	C9		RET		
051A	34FA13	OUT0:	LDA	13FAH	: CR0 TRIGGER CONTROL
051D	6F		MOV	L.A	
051E	087E		IN	7EH	: READ UP-DOWN SWITCH
0520	17		RAL		
0521	0A2B05		JC	0CHK	

0524	2C		INR	L	
0525	C32D05		JMP	RCHK	
0528	17	DCHK:	RAL		
0529	DA2D05		JC	RCHK	
052C	2D		DCR	L	
052D	7D	RCHK:	MOV	A+L	
052E	FEFF		CPI	0FFH	
0530	C23805		JNZ	RCH1	
0533	3E64		MVI	A+100	
0535	C93E05		JMP	RCH2	
0538	FE65	RCH1:	CPI	101	
053A	DA3E05		JC	RCH2	
053D	97		SUB	A	
053E	32FA13	RCH2:	STA	13FAH	
0541	D3EC		OUT	0ECH	: MS BYTE OF MULTIPLICAND
0543	3AFA13		LDA	13FAH	
0546	D3EE		OUT	0EEH	: MS BYTE OF MULTIPLIER
0548	97		SUB	A	
0549	D3ED		OUT	0EDH	: LS BYTE OF MULTIPLICAND
054B	D3DC		OUT	0DCH	: LS BYTE OF MULTIPLIER
054D	D8DD		IN	0DDH	: MS BYTE OF RESULT
054F	47		MOV	B+A	
0550	D8DE		IN	0DEH	: 2ND MS BYTE OF RESULT
0552	4F		MOV	C+A	
0553	110000		LXI	D+0	: CONVERSION FROM HEXADECIMAL
0556	2610		MVI	H+16	: TO DECIMAL
0558	79	L1:	MOV	A+C	
0559	17		RAL		
055A	4F		MOV	C+A	
055B	78		MOV	A+B	
055C	17		RAL		
055D	47		MOV	B+A	: CARRY = MS BIT OF BC
055E	78		MOV	A+E	
055F	86		ADC	E	: E = 2E + CARRY
0560	27		DAA		: CONVERT TO DECIMAL
0561	5F		MOV	E+A	
0562	7A		MOV	A+0	

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0563 8A      AOC      0
0564 27      DAA
0565 57      MOV      D+A      : D = 2D + CARRY
0566 25      DCR      H
0567 C25805  JNZ      L1
056A 78      MOV      A+E      : TO DIGITAL FREQUENCY DISPLAY
056B D37D     OUT      7DH
056D 7A      MOV      A,D
056E D37C     OUT      7CH
0570 2613     OUTP1:  MVI      H,13H      : DISPLAY SPECTRUM ON CRO
0572 0618     MVI      B,24      : DURATION = 1/8 SEC APPROX.
0574 3E02     MVI      A,2
0576 D3F6     OUT      0F6H      : PC1 = 1
0578 05      OUT1:   DCR      H
0579 08      RZ
057A 97      SUB      A
057B D3F6     OUT      0F6H      : TRIGGER CRO
057D 8F      MOV      L,A
057E 7E      OUT2:   MOV      A,M
057F D3F4     OUT      0F4H      : FREQUENCY COMPONENT TO CRO
0581 3AF413  LDA      13FAH     : FREQUENCY REFERENCE
0584 57      MOV      D,A
0585 7D      MOV      A,L
0586 8A      CMP      D
0587 3F      CMC
0588 3E00     MVI      A,0
05AA 17      RAL
05AB 17      RAL
05AC D3F6     OUT      0F6H      : POSITIVE EDGE INDICATES FREQUENCY
05AE 2C      INP      L
05AF 7D      MOV      A,L
0590 FE65     CPI      101
0592 CA7605  JZ      OUT1      : END OF FREQUENCY SCAN ?
0595 C37F05  JMP     OUT2      : NO. DISPLAY ANOTHER SPECTRAL POINT
059A 210013  REC:    LXI      H,1300H : RECORDER OUTPUT SUBROUTINE
059C 1F00     MVI      E,0
059E 1600     REC0:   MVI      D,0      : F IS MARKER PEN CONTROLLER
                                : 1 SECOND TIMER

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050F	08F6	TEST:	IN	0F6H	
0511	E610		ANI	10H	*SELECT PC4
0513	48		MOV	C+R	* FORMER IN C
0514	47		MOV	B+A	* LATTER IN C
0515	89		CMP	C	
0516	028F05		JNC	TEST	* DETECT NEGATIVE TRANSITION OF 256HZ
0519	14		INR	D	
051A	C24F05		JNZ	TEST	* 1 SECOND DELAY ELAPSED?
051D	3E04		MVI	A+4	
051E	03F6		OUT	0F6H	* SET MARKER TO 0
051I	7E		MOV	A+M	
0512	03F4		OUT	0F4H	* DATA TO D/A
0514	7R		MOV	A+E	
0515	8D		CMP	L	
0516	00C405		CZ	MARK	* IS MARKER REQUIRED?
0519	2C		INR	L	
051A	7D		MOV	A+L	
051E	FE65		CP1	101	
051D	C29D05		JNZ	REC0	* L.NE.101 ?
051D	97		SUB	A	
0511	03F6		OUT	0F6H	* ZERO O/P AND MARKER RELEASE
0513	76		HLT		
0514	C604	MARK:	ADI	10	
0516	5F		MOV	E+A	* E=F+10
0517	3E05		MVI	A+5	
0519	03F6		OUT	0F6H	* ENERGISE MARKER PEN
051E	09		RET		
051C	2E00	ENT:	MVI	L+0	*TEST STAIRCASE
051E	75	ENT1:	MOV	B+L	
051F	2C		INR	L	
051D	C20E05		JNZ	ENT1	
0513	0F		RST	1	
0514	2E00	AD30:	MVI	L+0	*TEST ADD 80H
0516	7E	ADH01:	MOV	A+M	
0517	C680		ADI	80H	
0519	77		MOV	B+A	
051A	2C		INR	L	

```

0508 C20605 JNZ A0801
050E 00 NOP
050F 7E TOUT: MOV A,M :TEST OUTPUT
0510 03FA OUT 0F4H
0512 2C INR L
0513 00 NOP
0514 00 NOP
0515 00 NOP
0516 00 NOP
0517 00 NOP
0518 00 NOP
0519 C20605 JNZ TOUT
051C 97 SHR A
051D 03FA OUT 0F6H
051F 3C02 MVI A,02H
0521 03FA OUT 0F6H
0523 C30605 JMP TOUT
0526 2F00 ENTA: MVI L,0 :TEST*ENTER CONSTANT DATA
0528 77 ENTA1: MOV M,A
0529 2C INR L
052A C2FA05 JNZ ENTA1
052C 0F RST 1

```

NO PROGRAM ERRORS

\* 01

A	0007	ACRE0	045C	ACRE1	045E	ACRE2	0492
A080	0504 *	A0801	0506	B	0000	C	0001
D	0002	DCHA	0526	E	0003	ENTA	05CC *
ENTA1	0508	ENTA	05F6 *	ENTA1	05F8	H	0004
L	0005	L1	0558	LOG	0484 *	LOG0	0486
LOG1	048F	LOG2	04D9	LOG3	050F	M	0006
MARK	0504	OUT0	0514 *	OUT1	0578	OUT2	057E
OUTPUT	0570 *	PSW	0006	RCH1	0538	RCH2	053E
RCHK	0520	REC	0598 *	REC0	0590	SP	0006
TEST	059F	TOUT	050F	WR	0417	WR1	0419
W32	042C	WC	0400 *	WH	0446	WH1	0448
W42	044A						

0600		DB	600H
0600	00	DB	00H
0601	03	DB	03H
0602	06	DB	06H
0603	09	DB	09H
0604	0C	DB	0CH
0605	0F	DB	0FH
0606	12	DB	12H
0607	15	DB	15H
0608	18	DB	18H
0609	1C	DB	1CH
060A	1F	DB	1FH
060B	22	DB	22H
060C	25	DB	25H
060D	28	DB	28H
060E	2B	DB	2BH
060F	2E	DB	2EH
0610	30	DB	30H
0611	33	DB	33H
0612	36	DB	36H
0613	39	DB	39H
0614	3C	DB	3CH
0615	3F	DB	3FH
0616	41	DB	41H
0617	44	DB	44H
0618	47	DB	47H
0619	49	DB	49H
061A	4C	DB	4CH
061B	4E	DB	4EH
061C	51	DB	51H
061D	53	DB	53H
061E	55	DB	55H
061F	58	DB	58H
0620	5A	DB	5AH
0621	5C	DB	5CH
0622	5E	DB	5EH
0623	60	DB	60H



0624	62	DB	62H
0625	64	DB	64H
0626	66	DB	66H
0627	68	DB	68H
0628	6A	DB	6AH
0629	6C	DB	6CH
062A	6D	DB	6DH
062B	6E	DB	6EH
062C	70	DB	70H
062D	72	DB	72H
062E	73	DB	73H
062F	75	DB	75H
0630	76	DB	76H
0631	77	DB	77H
0632	78	DB	78H
0633	79	DB	79H
0634	7A	DB	7AH
0635	7B	DB	7BH
0636	7C	DB	7CH
0637	7C	DB	7CH
0638	7D	DB	7DH
0639	7E	DB	7EH
063A	7F	DB	7FH
063B	7F	DB	7FH
063C	7F	DB	7FH
063D	7F	DB	7FH
063E	7F	DB	7FH
063F	7F	DB	7FH
0640	7F	DB	7FH
0641	7F	DB	7FH
0642	7F	DB	7FH
0643	7F	DB	7FH
0644	7F	DB	7FH
0645	7F	DB	7FH
0646	7F	DB	7FH
0647	7E	DB	7EH
0648	7D	DB	7DH

0649	7C	08	7CH
064A	7C	08	7CH
064B	7A	08	7BH
064C	7A	08	7AH
064D	79	08	79H
064E	78	08	78H
064F	77	08	77H
0650	76	08	76H
0651	75	08	75H
0652	73	08	73H
0653	72	08	72H
0654	70	08	70H
0655	6F	08	6FH
0656	6D	08	6DH
0657	6C	08	6CH
0658	6A	08	6AH
0659	6A	08	68H
065A	6A	08	66H
065B	64	08	64H
065C	62	08	62H
065D	60	08	60H
065E	5E	08	5EH
065F	5C	08	5CH
0660	5A	08	5AH
0661	58	08	58H
0662	55	08	55H
0663	53	08	53H
0664	51	08	51H
0665	4F	08	4EH
0666	4C	08	4CH
0667	49	08	49H
0668	47	08	47H
0669	44	08	44H
066A	41	08	41H
066B	3F	08	3FH
066C	3C	08	3CH
066D	39	08	39H

066F	36	DB	36H
066F	33	DB	33H
0670	30	DB	30H
0671	2F	DB	2FH
0672	2E	DB	2EH
0673	2B	DB	2BH
0674	25	DB	25H
0675	22	DB	22H
0676	1F	DB	1FH
0677	1C	DB	1CH
0678	18	DB	18H
0679	15	DB	15H
067A	12	DB	12H
067B	0F	DB	0FH
067C	0C	DB	0CH
067D	09	DB	09H
067E	06	DB	06H
067F	03	DB	03H
0680	00	DB	00H
0681	FC	DB	0FCH
0682	F9	DB	0F9H
0683	F6	DB	0F6H
0684	F3	DB	0F3H
0685	F0	DB	0F0H
0686	EF	DB	0EFH
0687	EA	DB	0EAH
0688	E7	DB	0E7H
0689	E3	DB	0E3H
068A	E0	DB	0E0H
068B	DD	DB	0DDH
068C	DA	DB	0DAH
068D	D7	DB	0D7H
068E	D4	DB	0D4H
068F	D1	DB	0D1H
0690	CE	DB	0CEH
0691	CB	DB	0CBH
0692	C9	DB	0C9H

0693	C6	DR	0C6H
0694	C3	DR	0C3H
0695	C0	DR	0C0H
0696	B8	DR	0B8H
0697	B5	DR	0B5H
0698	9A	DR	09AH
0699	B6	DR	0B6H
069A	B3	DR	0B3H
069B	B1	DR	0B1H
069C	AE	DR	0AEH
069D	AC	DR	0ACH
069E	AA	DR	0AAH
069F	A7	DR	0A7H
06A0	A5	DR	0A5H
06A1	A3	DR	0A3H
06A2	A1	DR	0A1H
06A3	9F	DR	09FH
06A4	9D	DR	09DH
06A5	9E	DR	09EH
06A6	99	DR	099H
06A7	97	DR	097H
06A8	95	DR	095H
06A9	93	DR	093H
06AA	92	DR	092H
06AB	90	DR	090H
06AC	8F	DR	08FH
06AD	8D	DR	08DH
06AE	8C	DR	08CH
06AF	8A	DR	08AH
06B0	89	DR	089H
06B1	88	DR	088H
06B2	87	DR	087H
06B3	86	DR	086H
06B4	85	DR	085H
06B5	84	DR	084H
06B6	83	DR	083H
06B7	83	DR	083H

06B8	82	DB	82H
06B9	81	DB	81H
06BA	81	DB	81H
06BB	80	DB	80H
06BC	80	DB	80H
06BD	80	DB	80H
06BE	80	DB	80H
06BF	80	DB	80H
06C0	80	DB	80H
06C1	80	DB	80H
06C2	80	DB	80H
06C3	80	DB	80H
06C4	80	DB	80H
06C5	80	DB	80H
06C6	81	DB	81H
06C7	81	DB	81H
06C8	82	DB	82H
06C9	83	DB	83H
06CA	83	DB	83H
06CB	84	DB	84H
06CC	85	DB	85H
06CD	86	DB	86H
06CE	87	DB	87H
06CF	88	DB	88H
06D0	89	DB	89H
06D1	8A	DB	8AH
06D2	8C	DB	8CH
06D3	8D	DB	8DH
06D4	8E	DB	8EH
06D5	90	DB	90H
06D6	92	DB	92H
06D7	93	DB	93H
06D8	95	DB	95H
06D9	97	DB	97H
06DA	99	DB	99H
06DB	9B	DB	9BH
06DC	9D	DB	9DH

06DF	9F	DR	9FH
06DF	A1	DR	0A1H
06DF	A3	DR	0A3H
06E0	A5	DR	0A5H
06E1	A7	DR	0A7H
06E2	A4	DR	0A4H
06E3	AC	DR	0ACH
06E4	AF	DR	0AEH
06E5	B1	DR	0B1H
06E6	B3	DR	0B3H
06E7	B6	DR	0B6H
06E8	B8	DR	0B8H
06E9	BB	DR	0BBH
06EA	BE	DR	0BEH
06EB	C0	DR	0C0H
06EC	C3	DR	0C3H
06ED	C6	DR	0C6H
06EE	C9	DR	0C9H
06EF	CC	DR	0CCH
06F0	CF	DR	0CFH
06F1	D1	DR	0D1H
06F2	D4	DR	0D4H
06F3	D7	DR	0D7H
06F4	DA	DR	0DAH
06F5	DD	DR	0DDH
06F6	E0	DR	0E0H
06F7	E3	DR	0E3H
06F8	E7	DR	0E7H
06F9	EA	DR	0EAH
06FA	ED	DR	0EDH
06FB	F0	DR	0F0H
06FC	F3	DR	0F3H
06FD	F6	DR	0F6H
06FE	F9	DR	0F9H
06FF	FC	DR	0FCH
0700	00	DR	00H
0701	24	DR	24H

0702	48	DB	48H
0703	68	DB	68H
0704	8C	DB	8CH
0705	A8	DB	0A8H
0706	CA	DB	0CAH
0707	E2	DB	0E2H
0708	F9	DB	0F9H
0709	0C	DB	0CH
070A	1A	DB	1AH
070B	24	DB	24H
070C	28	DB	28H
070D	27	DB	27H
070E	1F	DB	1FH
070F	11	DB	11H
0710	FC	DB	0FCH
0711	0F	DB	0DFH
0712	8A	DB	08AH
0713	8D	DB	8DH
0714	57	DB	57H
0715	17	DB	17H
0716	CE	DB	0CEH
0717	78	DB	78H
0718	1D	DB	1DH
0719	84	DB	084H
071A	40	DB	40H
071B	C0	DB	0C0H
071C	34	DB	34H
071D	98	DB	98H
071E	F6	DB	0F6H
071F	43	DB	43H
0720	82	DB	82H
0721	H4	DB	084H
0722	07	DB	007H
0723	EC	DB	0ECH
0724	F2	DB	0F2H
0725	E9	DB	0E9H
0726	00	DB	000H

0727	A7	DR	0A7H
0728	6F	DR	6EH
0729	24	DR	24H
072A	CA	DR	0CAH
072B	5F	DR	5FH
072C	E3	DR	0E3H
072D	5B	DR	5BH
072E	B6	DR	0B6H
072F	05	DR	05H
0730	42	DR	42H
0731	6C	DR	6CH
0732	85	DR	85H
0733	8A	DR	8AH
0734	7D	DR	7DH
0735	5D	DR	5DH
0736	2A	DR	2AH
0737	E4	DR	0E4H
0738	84	DR	84H
0739	1E	DR	1EH
073A	9D	DR	9DH
073B	0A	DR	0AH
073C	62	DR	62H
073D	A7	DR	0A7H
073E	D9	DR	0D9H
073F	F6	DR	0F6H
0740	FF	DR	0FFH
0741	F6	DR	0F6H
0742	D9	DR	0D9H
0743	A7	DR	0A7H
0744	62	DR	62H
0745	0A	DR	0AH
0746	9D	DR	9DH
0747	1E	DR	1EH
0748	84	DR	84H
0749	E4	DR	0E4H
074A	2A	DR	2AH
074B	5D	DR	5DH



074C	7D	DR	7DH
074D	84	DR	84H
074E	85	DR	85H
074F	8C	DR	8CH
0750	42	DR	42H
0751	05	DR	05H
0752	86	DR	086H
0753	55	DR	55H
0754	E3	DR	0E3H
0755	5F	DR	5FH
0756	CA	DR	0CAH
0757	24	DR	24H
0758	6E	DR	6EH
0759	A7	DR	0A7H
075A	D0	DR	0D0H
075B	E9	DR	0E9H
075C	F2	DR	0F2H
075D	FC	DR	0FCH
075E	07	DR	007H
075F	84	DR	084H
0760	82	DR	82H
0761	43	DR	43H
0762	F6	DR	0F6H
0763	98	DR	98H
0764	34	DR	34H
0765	C0	DR	0C0H
0766	40	DR	40H
0767	84	DR	084H
0768	10	DR	10H
0769	75	DR	75H
076A	CF	DR	0CFH
076B	17	DR	17H
076C	57	DR	57H
076D	80	DR	80H
076E	BA	DR	0BAH
076F	DF	DR	0DFH
0770	FC	DR	0FCH

0771	11	0A	11H
0772	1F	0B	1FH
0773	27	0C	27H
0774	28	0D	28H
0775	24	0E	24H
0776	1A	0F	1AH
0777	0C	10	0CH
0778	F9	11	0F9H
0779	E2	12	0E2H
077A	C8	13	0C8H
077B	4A	14	04AH
077C	8C	15	08CH
077D	64	16	064H
077E	45	17	045H
077F	24	18	024H
0780	00	19	00H
0781	DC	1A	0DCH
0782	6A	1B	06AH
0783	95	1C	095H
0784	74	1D	074H
0785	55	1E	055H
0786	38	1F	038H
0787	1E	20	01EH
0788	07	21	007H
0789	F4	22	0F4H
078A	F6	23	0F6H
078B	0C	24	00CH
078C	D5	25	0D5H
078D	C9	26	0C9H
078E	E1	27	0E1H
078F	EF	28	0EFH
0790	04	29	004H
0791	21	2A	021H
0792	46	2B	046H
0793	73	2C	073H
0794	89	2D	089H
0795	F9	2E	0F9H

0796	32	DR	32H
0797	85	DR	85H
0798	E3	DR	0E3H
0799	4C	DR	4CH
079A	C8	DR	0C8H
079B	40	DR	40H
079C	CC	DR	0CCH
079D	65	DR	65H
079E	0A	DR	0AH
079F	8D	DR	08DH
07A0	7F	DR	7FH
07A1	4C	DR	4CH
07A2	29	DR	29H
07A3	14	DR	14H
07A4	0E	DR	0EH
07A5	17	DR	17H
07A6	30	DR	30H
07A7	59	DR	59H
07A8	92	DR	92H
07A9	DC	DR	0DCH
07AA	38	DR	38H
07AB	41	DR	041H
07AC	1D	DR	1DH
07AD	4R	DR	04RH
07AE	4A	DR	4AH
07AF	FB	DR	0FBH
07B0	BE	DR	0BEH
07B1	94	DR	94H
07B2	7K	DR	7KH
07B3	76	DR	76H
07B4	83	DR	83H
07B5	43	DR	043H
07B6	06	DR	006H
07B7	1C	DR	1CH
07B8	76	DR	76H
07B9	E2	DR	0E2H
07BA	63	DR	63H

0758	F6	08	0F6H
075C	9E	08	9EH
0760	59	08	59H
076E	27	08	27H
07BF	0A	08	0AH
07C0	01	08	01H
07C1	0A	08	0AH
07C2	27	08	27H
07C3	59	08	59H
07C4	9F	08	9EH
07C5	F6	08	0F6H
07C6	63	08	63H
07C7	E2	08	0E2H
07C8	76	08	76H
07C9	1C	08	1CH
07CA	D6	08	0D6H
07CB	A3	08	0A3H
07CC	83	08	83H
07CD	76	08	76H
07CE	74	08	74H
07CF	94	08	94H
07D0	6E	08	06EH
07D1	FA	08	0FAH
07D2	4A	08	4AH
07D3	48	08	048H
07D4	1C	08	1CH
07D5	A1	08	0A1H
07D6	36	08	36H
07D7	DC	08	0DCH
07D8	92	08	92H
07D9	59	08	59H
07DA	30	08	30H
07DB	17	08	17H
07DC	8E	08	0EH
07DD	14	08	14H
07DE	29	08	29H
07DF	4C	08	4CH

07F0	7E	DB	7EH
07F1	8D	DB	080H
07F2	0A	DB	0AH
07F3	65	DB	65H
07F4	CC	DB	0CCH
07F5	40	DB	40H
07F6	00	DB	0C0H
07F7	4C	DB	4CH
07F8	E3	DB	0E3H
07F9	85	DB	85H
07FA	32	DB	32H
07FB	F9	DB	0E9H
07FC	A9	DB	0A9H
07FD	73	DB	73H
07FE	46	DB	46H
07FF	21	DB	21H
07F0	04	DB	04H
07F1	EF	DB	0EFH
07F2	E1	DB	0E1H
07F3	09	DB	009H
07F4	08	DB	008H
07F5	0C	DB	00CH
07F6	E6	DB	0E6H
07F7	F4	DB	0F4H
07F8	07	DB	07H
07F9	1E	DB	1EH
07FA	38	DB	38H
07FB	55	DB	55H
07FC	74	DB	74H
07FD	95	DB	95H
07FE	B8	DB	0B8H
07FF	DC	DB	0DCH

NO PROGRAM ERRORS

\* 01

A	0007	H	0000	C	0001	D	0002
E	0003	H	0004	L	0005	M	0006
PSW	0006	SP	0006				

APPENDIX 3

THE DFT OF THE STEP FUNCTION

The DFT as used in the spectrum analyser is defined as :

$$X_n = (1/N) \sum_{k=0}^{N-1} x_k \exp(-j2\pi nk/N)$$

Consider the input signal with the following properties :

$$x_k = \begin{cases} 1 & \text{for } k = 0 \dots (N/2-1) \\ 0 & \text{for } k = (N/2) \dots (N-1) \end{cases}$$

$$\therefore X_n = (1/N) \sum_{k=0}^{(N/2-1)} \exp(-j\pi nk/N)$$

$$\therefore X_0 = (1/N) \sum_{k=0}^{(N/2-1)} 1 = (1/N) \cdot (N/2)$$

$$\therefore X_0 = 0.5$$

for  $n > 0$

$$X_n = (1 - \exp(-j\pi n)) / (1 - \exp(-j2\pi n/N)) / N$$

If  $n$  is even  $\exp(-j\pi n) = 1$

$$\therefore X_n = 0$$

If  $n$  is odd  $\exp(-j\pi n) = -1$

$$\therefore X_n = (2/N) / (1 - \exp(-j2\pi n/N))$$

$$= \frac{(2/N)}{\exp(-j\pi n/N)(\exp(j\pi n/N) - \exp(-j\pi n/N))}$$

$$= (-j/N \exp(j\pi n/N) / \sin(\pi n/N))$$

$$\therefore X_n = (1/N) - j (1/N) \cos(\pi n/N) / \sin(\pi n/N)$$

for odd  $n$