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THE DESIGN AND EXPERIMENTAL INVESTIGATION  
OF A HYBRID PHASE-LOCK LOOP

BY

SRINIVASA H.R. RAGHAVAN, 1944-

A  
THESIS

Presented to the Faculty of the Graduate School of the  
UNIVERSITY OF MISSOURI-ROLLA

In Partial Fulfillment of the Requirements for the Degree  
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1971

Approved by

R. E. Ziemer (Advisor)

William H. Frank

Max Engelhardt

## ABSTRACT

This thesis describes the design and the laboratory implementation of a hybrid phase-lock loop. The step-by-step procedure followed in developing this experimental hybrid phase-lock loop, and its optimization with respect to various parameters are given. In regard to the performance characteristics, the loop was used to obtain the phase estimate of a split-phase modulated signal. The phase error probability density function was measured under various input noise conditions and these results are included and compared with previously published theoretical results.

## TABLE OF CONTENTS

	Page
ABSTRACT . . . . .	ii
LIST OF FIGURES . . . . .	iv
LIST OF TABLES . . . . .	vi
ACKNOWLEDGEMENTS . . . . .	vii
I. INTRODUCTION . . . . .	1
II. DESCRIPTION OF THE HYBRID LOOP AND EQUATIONS OF OPERATION . . . . .	3
III. DESIGN OF THE HYBRID LOOP . . . . .	10
IV. RESULTS AND DISCUSSIONS . . . . .	13
V. CONCLUSIONS . . . . .	27
REFERENCES . . . . .	28
VITA . . . . .	30
APPENDICES	
A. EXPERIMENTAL HYBRID LOOP COMPONENTS . . . . .	31
B. DESIGN EXAMPLE . . . . .	39
C. PHASE ERROR PROBABILITY DENSITY MEASUREMENT.	43

## LIST OF FIGURES

Figure		Page
1.	Block Diagram of a Hybrid Phase-Lock Loop . . . . .	4
2.	Experimental Hybrid Phase-Lock Loop . . . . .	12
3.	Modulating PN Sequence and Demodulated Output Sequence . . . . .	14
4.	Probability Density Function of Phase Error; $B_L=85$ Hz; $\rho=0.3$ , $m=1/2$ , $SNR=25$ dB . . . . .	15
5.	Probability Density Function of Phase Error; $B_L=85$ Hz; $\rho=0.3$ , $m=1/2$ , $SNR=19$ dB . . . . .	16
6.	Probability Density Function of Phase Error; $B_L=85$ Hz; $\rho=0.3$ , $m=1/2$ , $SNR=17$ dB . . . . .	17
7.	Probability Density Function of Phase Error; $B_L=85$ Hz; $\rho=0.3$ , $m=1$ , $SNR=25$ dB . . . . .	18
8.	Probability Density Function of Phase Error; $B_L=85$ Hz; $\rho=0.3$ , $m=1$ , $SNR=19$ dB . . . . .	19
9.	Probability Density Function of Phase Error; $B_L=85$ Hz; $\rho=0.3$ , $m=1$ , $SNR=17$ dB . . . . .	20
10.	Probability Density Function of Phase Error; $B_L=85$ Hz, $\rho=0.3$ , $m=0.2$ , $SNR=25$ dB . . . . .	21
11.	Voltage Controlled Oscillator . . . . .	32
12.	VCO Characteristic . . . . .	33
13.	Phase Detectors . . . . .	35

## LIST OF FIGURES (continued)

Figure		Page
14.	Outer Loop Filters . . . . .	36
15.	Adder, Loop Filter, and Inverter . . . . .	37
16.	Experimental Set Up for Measurement of Phase Error Probability Density . . . . .	45
17.	Exclusive-OR Phase Detector . . . . .	46
18.	Phase Detector Characteristic . . . . .	47

## LIST OF TABLES

Table		Page
I	Tabulation of Loop Phase Error Under Various Conditions . . . . .	23
II	Summary of Loop Component Values . . . . .	42

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## I. INTRODUCTION

The hybrid phase-lock loop (PLL) is a combination of a Costas loop [1]\* and a phase-lock loop [2]. The Costas loop portion responds to the modulation component and the phase-lock loop portion responds to the carrier component of the received signal. In a hybrid loop the phase estimate for coherent detection is derived both from the carrier and data components. The superior performances of the phase-lock loop and Costas loop at low and high signal-to-noise ratios (SNRs), respectively, are therefore combined to obtain a better performance for all SNRs than for a Costas or phase-lock loop alone.

This thesis is concerned with the development of an experimental implementation of a hybrid phase-lock loop which is optimized for minimum mean-square phase error with respect to its design parameters. The design procedure is considered in detail. Also performance is characterized in terms of mean-square phase error versus input SNR.

The hybrid phase-lock loop is presently of interest due to the possibility that it may perform better than conventional tracking loops in non-Gaussian and non-stationary environments such as radio-frequency interference, multipath and fading. Such backgrounds may be severe problems in various synchronous-orbit relay-

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\*Numbers in [] denote references

satellite systems now under study [3]. Also of interest is possible application in low data-rate systems [4]. Theoretical studies have been made of the performance of a hybrid phase-lock loop as compared to the performance of a Costas or a conventional phase-lock loop [4,5,6]. The well-known Fokker Planck techniques were used in these studies, and a maximum improvement of 1.5dB in SNR was predicted. The optimization of the hybrid loop with respect to its design parameters has been dealt with in detail in [5]. This approach has been used in this study to optimize the experimental model of the hybrid loop developed and will be described in the latter sections. Before doing so, however, the hybrid loop will be described in more detail and its equations of operation developed.

## II. DESCRIPTION OF THE HYBRID LOOP AND EQUATIONS OF OPERATION

A block diagram of a hybrid loop is shown in Fig. 1. As previously mentioned it is essentially a combination of a Costas loop and a phase-lock loop. The outer loop (ABCDEFGHI), referred to as the Costas loop portion, is designed to lock on the modulation component of the input signal. The inner loop (ABGHI), referred to as phase-lock loop portion, is designed to respond to the carrier component of the input signal. The error signals from both branches of the loop are added and passed through the loop filter  $F(p)$ . The output of the loop filter is used as the control voltage of the voltage-controlled oscillator (VCO). The filters with transfer functions  $G(p)$  of the outer legs remove the double frequency components resulting from the multiplication operations as well as the low-frequency component due to the carrier component at the input. To permit removal of this low frequency component, the data input should have negligible energy in its spectrum close to the origin; this is true, for example, if split phase modulation is used for transmission of the data, or if the data is placed on a sub-carrier.

The equations of operation are easily derived [5]. In Fig. 1, let

$$y(t) = s(t) + n(t) \quad (1)$$

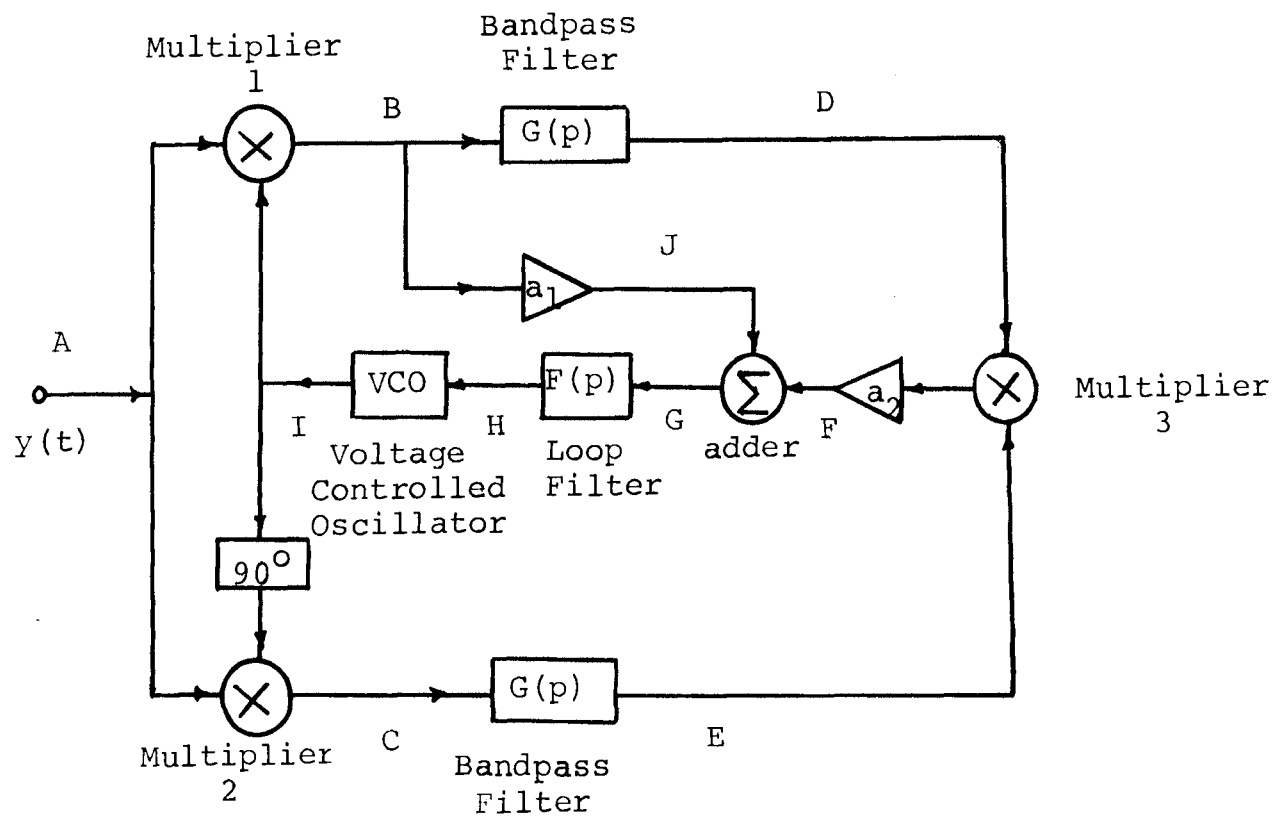


Figure 1. Block Diagram of a Hybrid Phase-Lock Loop

where

$$s(t) = \sqrt{2} A \text{Sin}(\omega t + x(t) \text{Cos}^{-1} m + \theta) \quad (2)$$

is the signal, and

$$n(t) = \sqrt{2} n_c(t) \text{Cos}(\omega t + \theta) + \sqrt{2} n_s(t) \text{Sin}(\omega t + \theta) \quad (3)$$

represents a Gaussian noise process. In (2),  $x(t) = \pm 1$  in  $T_B$  - second intervals is the modulation and  $m$  is the square-root of the ratio of the power in the reference signal to the total transmitted power. The signal component of the input,  $s(t)$ , can also be written as

$$s(t) = \sqrt{2} A m \text{Sin}(\omega t + \theta) + \sqrt{2} A \sqrt{1-m^2} x(t) \text{Cos}(\omega t + \theta)$$

where the first term is the carrier component and the second term is the modulation component.

The VCO output is assumed to be

$$\sqrt{2} k_1 \text{Cos}(\omega t + \hat{\theta}) .$$

This yields

$$\begin{aligned} y_B(t) = k_o [ & 2k_1 A m \text{Sin}(\omega t + \theta) \text{Cos}(\omega t + \hat{\theta}) + \\ & 2A k_1 \sqrt{1-m^2} x(t) \text{Cos}(\omega t + \theta) \text{Cos}(\omega t + \hat{\theta}) + \\ & \sqrt{2} k_1 \text{Cos}(\omega t + \hat{\theta}) n(t) ] \quad (4) \end{aligned}$$

for the output of Multiplier 1. Similarly, the output of Multiplier 2 is

$$\begin{aligned}
y_C(t) = k_0 [ & 2 k_1 A m \sin(\omega t + \theta) \sin(\omega t + \hat{\theta}) + \\
& 2 A k_1 \sqrt{1-m^2} x(t) \cos(\omega t + \theta) \sin(\omega t + \hat{\theta}) + \\
& \sqrt{2} k_1 \sin(\omega t + \hat{\theta}) n(t) ].
\end{aligned} \tag{5}$$

The filters with transfer function  $G(p)$  eliminate the double frequency and dc components. These signals,  $y_D(t)$  and  $y_E(t)$  are multiplied in multiplier M3 producing the outer loop control voltage

$$\begin{aligned}
e_C(t) = k_0^2 k_2 [ & \frac{A^2 (1-m^2)}{2} ] k_1^2 \sin 2\vartheta + A k_1 \sqrt{1-m^2} x(t) n_C(t). \\
& \sin 2\vartheta - k_1 A \sqrt{1-m^2} x(t) n_S(t) \cos 2\vartheta + \\
& \frac{k_1^2 n_C^2}{2} \sin 2\vartheta - k_1^2 n_C n_S \cos 2\vartheta - \frac{n_S^2 k_1^2}{2} \sin 2\vartheta ],
\end{aligned} \tag{6}$$

where  $\vartheta = \theta - \hat{\theta}$  is the phase error.

The control voltage for the phase-lock loop portion is

$$e_P(t) = k_1 A m \sin \vartheta + k_1 n_C(t) \cos \vartheta + k_1 n_S(t) \sin \vartheta, \tag{7}$$

and the total control voltage due to outer and inner loops is

$$e_H(t) = F(p) [a_1 e_P(t) + a_2 e_C(t)].$$

Here a transfer function

$$F(p) = \frac{p+a}{p+\varepsilon} \tag{8}$$

is used, where  $a$  and  $\epsilon$  are constants. This corresponds to the case of an imperfect second-order loop. The differential equation for the loop can be written as

$$\frac{d\hat{\theta}}{dt} = +k_v F(p) [a_1 e_p(t) + a_2 e_c(t)] \quad (9)$$

or

$$\frac{d\theta}{dt} - \frac{d\vartheta}{dt} = +k_v F(p) [a_1 e_p(t) + a_2 e_c(t)]. \quad (10)$$

Considering only the signal portion of  $e_p(t)$  and  $e_c(t)$ , linearizing the equation by assuming  $\vartheta$  to be small, substituting  $F(p) = \frac{p+a}{p+\epsilon}$ , and simplifying, yields

$$\begin{aligned} \frac{d^2\vartheta}{dt^2} + \{k_v k_o k_1 A [m + \frac{a_2}{a_1} k_o k_2 k_1 A (1-m^2)] + \epsilon\} \frac{d\vartheta}{dt} + \\ k_v k_o k_1 A a [m + \frac{a_2}{a_1} k_o k_2 k_1 A (1-m^2)] \vartheta(t) \\ = \frac{d^2\theta}{dt^2} + \epsilon \frac{d\theta}{dt}. \end{aligned} \quad (11)$$

Letting

$$G_o = k_v k_o k_1 A [m + \frac{a_2}{a_1} k_o k_2 k_1 A (1-m^2)], \quad (12)$$

$$\epsilon + G_o = 2\xi\omega_n, \quad (13)$$

and

$$\omega_n^2 = G_o a \quad (14)$$

allows (11) to be written as

$$\ddot{\vartheta}(t) + 2\xi\omega_n\dot{\vartheta}(t) + \omega_n^2\vartheta(t) = \ddot{\theta}(t) + \varepsilon\dot{\theta}(t). \quad (15)$$

This is a differential equation for a second-order linear system written in terms of the parameters  $\xi$ , the damping factor, and  $\omega_n$ , the natural frequency. Together with the imperfection factor  $\varepsilon$ ,  $\omega_n$  and  $\xi$  completely characterize the loop response to an arbitrary input,  $\theta(t)$ , provided that the input is small so that linearization holds. The nonlinear equation of loop operation, (10), can also be expressed in terms of these parameters.

Quite often it is convenient to characterize the loop in terms of its single-sided equivalent noise bandwidth  $B_L$ , which can be obtained from

$$B_L = \frac{1}{H^2(0)} \int_0^{\infty} |H(j\omega)|^2 \frac{d\omega}{2\pi}, \quad (16)$$

where  $H(j\omega)$ , the closed loop transfer function is

$$H(s) = \frac{\hat{\theta}(s)}{\theta(s)} = 1 - \frac{\vartheta(s)}{\theta(s)} \quad (17)$$

From (15) we obtain

$$H(s) = \frac{G_o s + aG_o}{s^2 + (\varepsilon + G_o)s + aG_o}. \quad (18)$$

Carrying out the integration for  $B_L$  yields

$$B_L = \frac{G_o + a}{4(1 + \varepsilon/G_o)}. \quad (19)$$



Design of the loop consists of choosing appropriate values for  $G_0$ ,  $a$  and  $\epsilon$  to get the required bandwidth and damping factor. The procedure is outlined in the next section.

### III. DESIGN OF THE HYBRID LOOP

The design procedure is based on the theory outlined in the previous section. The design equations for the hybrid loop have been shown to be:

$$1. \text{ Loop Bandwidth; } B_L = \frac{G_o + a}{4(1 + \epsilon/G_o)} \quad (20)$$

$$2. \text{ Loop Damping Factor; } \xi = \frac{1}{2} \cdot \frac{\epsilon + G_o}{\sqrt{aG_o}} \quad (21)$$

The value of  $\xi$  was chosen to be  $1/\sqrt{2}$ . Jaffe and Rehtin [7] have shown that this choice of  $\xi$  minimizes phase error variance due to additive noise at the input, while maintaining a specified maximum transient phase error due to the step changes in input phase. Once the value of  $\xi$  is fixed at  $1/\sqrt{2}$ , the expression for loop bandwidth can be written as

$$B_L = G_o \cdot \frac{1 + a/G_o}{4\sqrt{2a/G_o}} \quad (22)$$

where  $a$  is the loop filter time constant and  $\epsilon$  is the loop filter imperfection factor. Also,

$$G_o = k_v k_3 (a_1 k_o) (k_1 A) [m + \rho(1 - m^2)] \quad (23)$$

where

$k_v$  is VCO constant (rad/sec./volt),

$k_3$  is the loop filter gain factor,

$a_1$  is inner loop gain,

$$m = \frac{\text{Power in the reference } (P_V)}{\text{Total transmitted power } (P_T)} \quad (24)$$

and

$$\rho = \frac{a_2}{a_1} k_2 k_o k_1 A. \quad (25)$$

For a given value of loop bandwidth and  $\xi$  of  $1/\sqrt{2}$ , it remains to optimize the loop with respect to  $m$  and  $\rho$ . Computerized results showing the values of  $\rho_{opt.}$  and  $m_{opt.}$  for different values of input SNR are given in [5]. The loop filter time constant,  $a$ , and the loop imperfection factor,  $\epsilon$ , are determined by solving the two equations giving  $B_L$  and  $\xi$  in terms of loop parameters.

The hybrid loop design is shown in Fig. 2. Detailed design of the various loop components are given in Appendix A, and an illustrative design example is given in Appendix B. A summary of the values of the various parameters pertinent to the design shown in Fig. 2 is given in Table II in Appendix B. For  $\xi = .707$ , loop bandwidths of 8.5, 85, 850 and 8500 Hz, are achieved by switching various resistors and capacitors into the feedback loops of the operational amplifiers. By adjusting the potentiometers at the summing junction of the first operational amplifier  $\rho$  can be adjusted from 0.05 to 2.5.

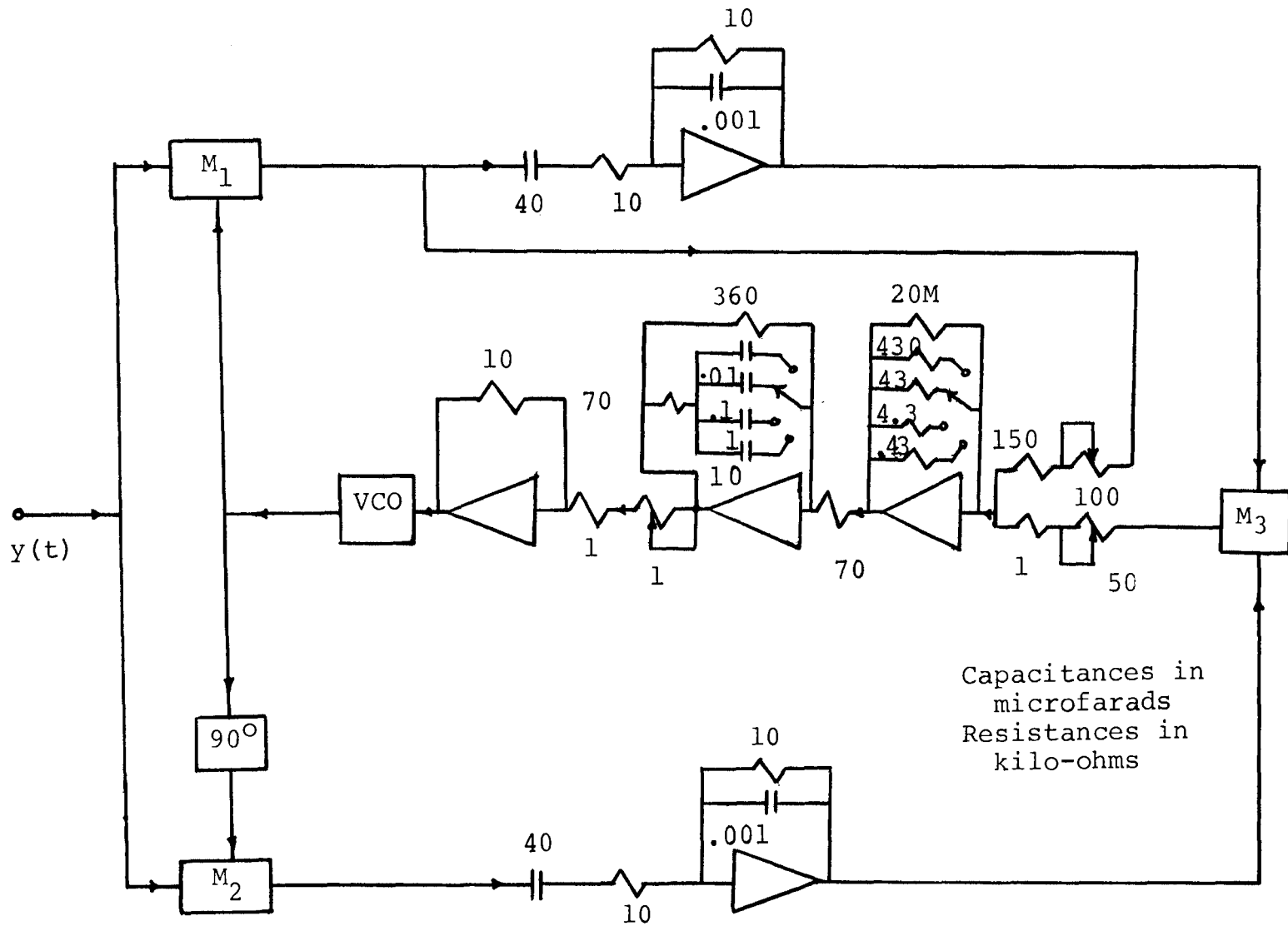


Figure 2. Experimental Hybrid Phase-Lock Loop

#### IV. RESULTS AND DISCUSSIONS

Performance of the loop was experimentally tested by demodulating a phase modulated signal in the presence of noise, and measuring the phase error probability density function. Details of the experimental set up are given in Appendix C.

In Fig. 3, the modulating PN-sequence and the demodulated PN-sequence, when no external noise was added to the input signal, are shown. The modulating signal is a 63 bit maximal-length PN code with a bit rate of 11.7k Hz.

The input to the loop is a split-phase modulated signal, i.e. a logic one is encoded as a plus-to-minus phase transition and a logic zero as a minus-to-plus transition, or vice-versa.

Phase error probability density functions for various loop and noise conditions are shown in Figures 4-10. The phase error standard deviations obtained from these phase error density plots at different SNRs and for various modulation indices are tabulated in Table I. The last column in Table I represents the value of the phase-error standard deviation of the hybrid loop ( $\sigma_H$ ) calculated theoretically from the results of [5]. From [5]

$$\sigma_H^2 = 1 + \frac{\rho^2(1-m^2) \left[ 1 + \frac{1}{z\beta(1-m^2)} \right]}{z[m + \rho(1-m^2)]} \quad (26)$$

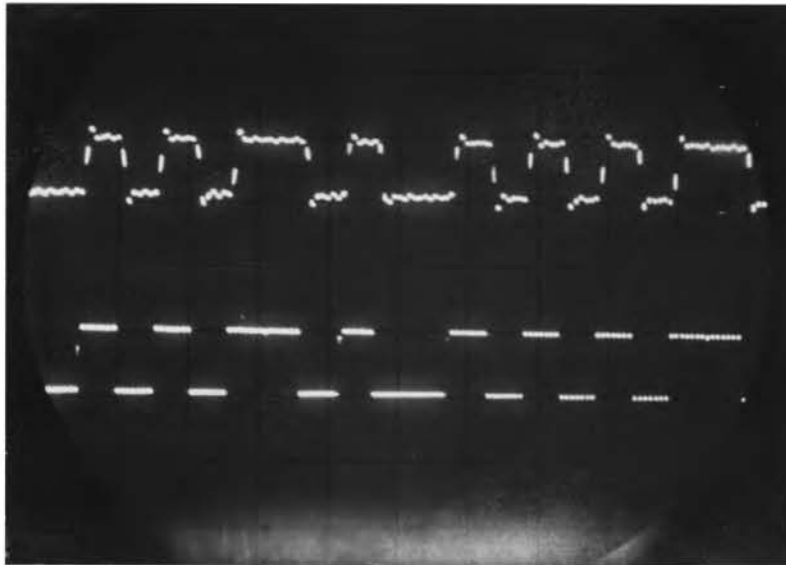
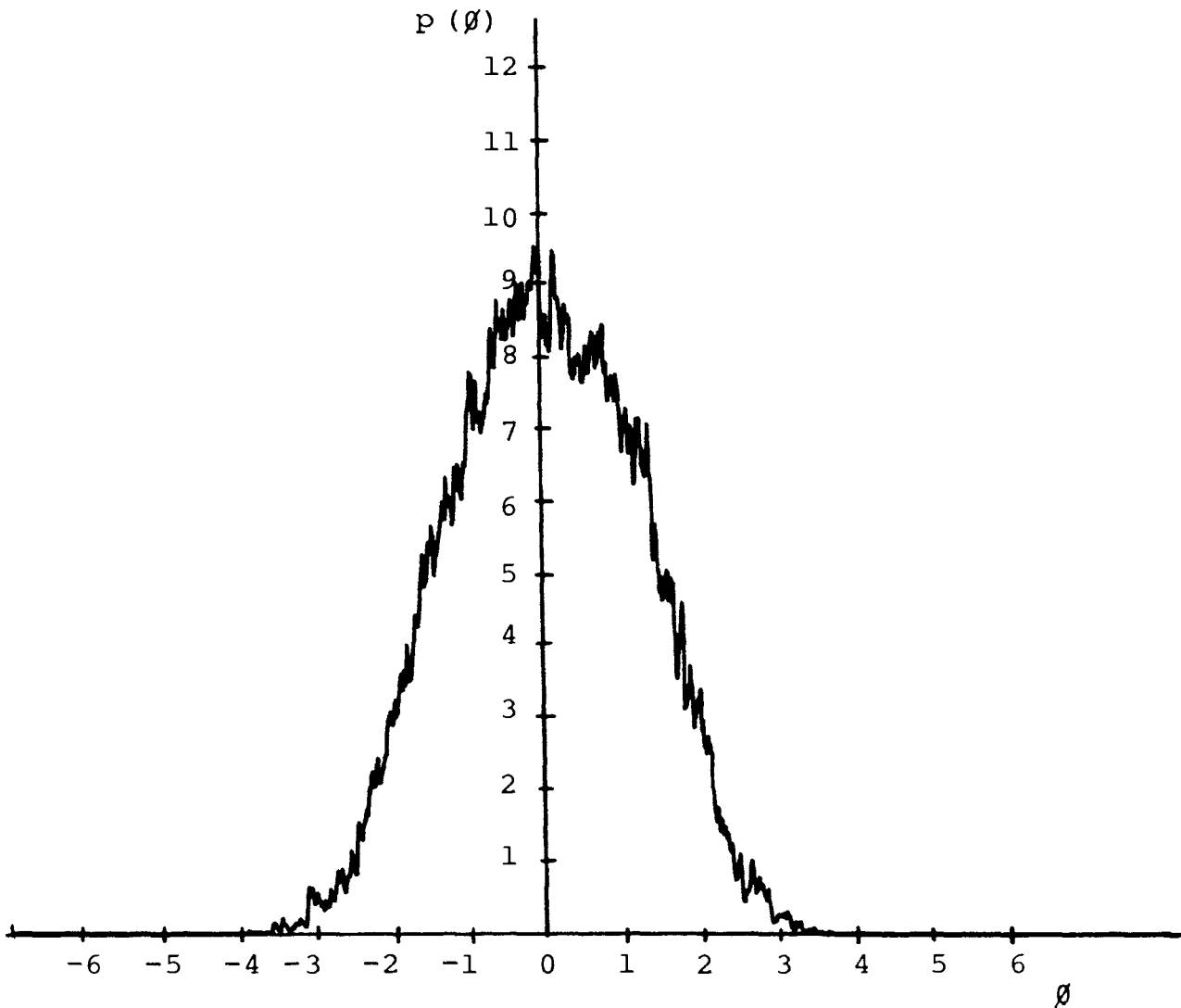
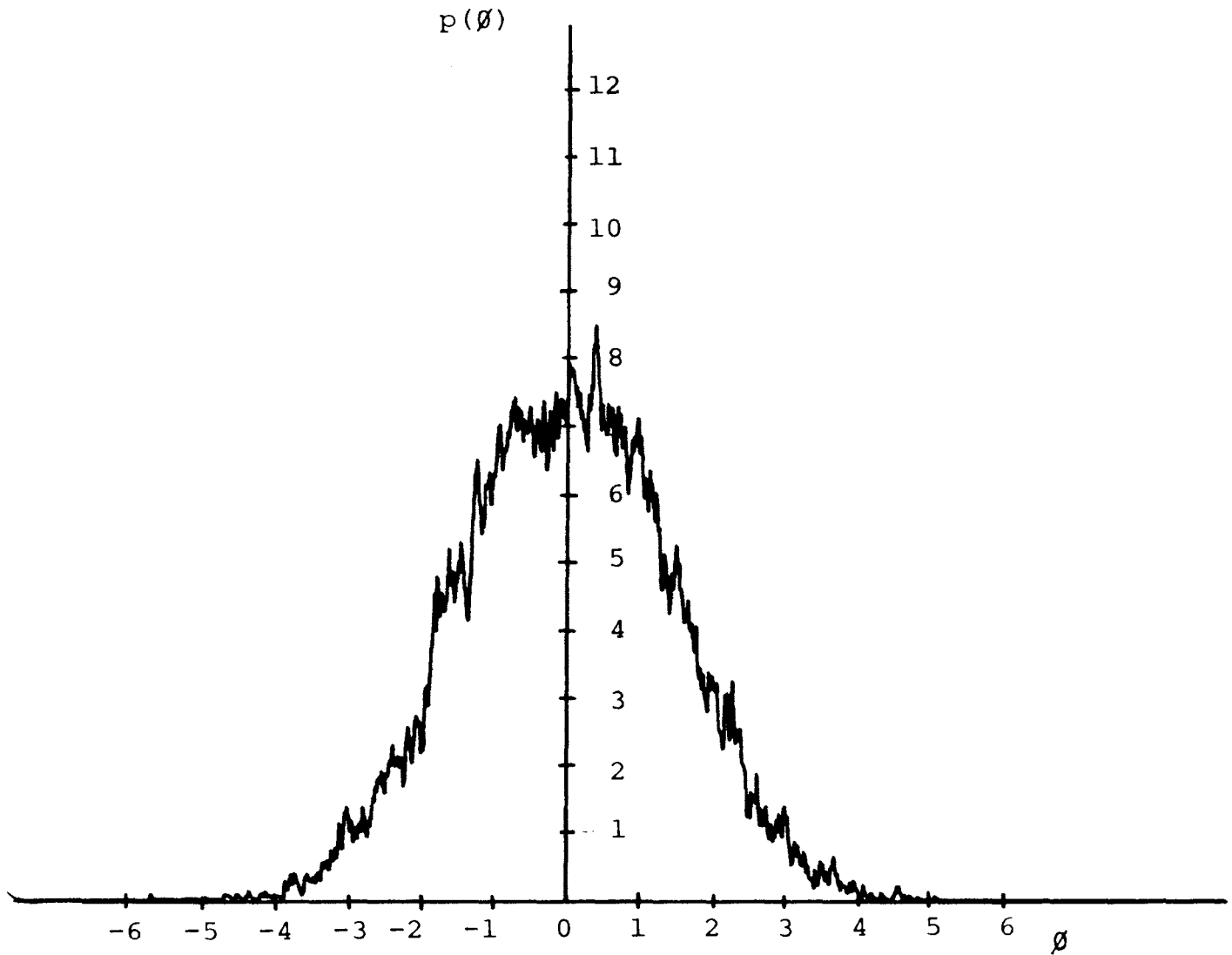


Figure 3. Modulating PN Sequence (lower trace) and Demodulated Output Sequence (upper trace)



Calibration: Abscissa 1 unit = 1.12 radians  
Ordinate 1 unit = .0377 rad.<sup>-1</sup>

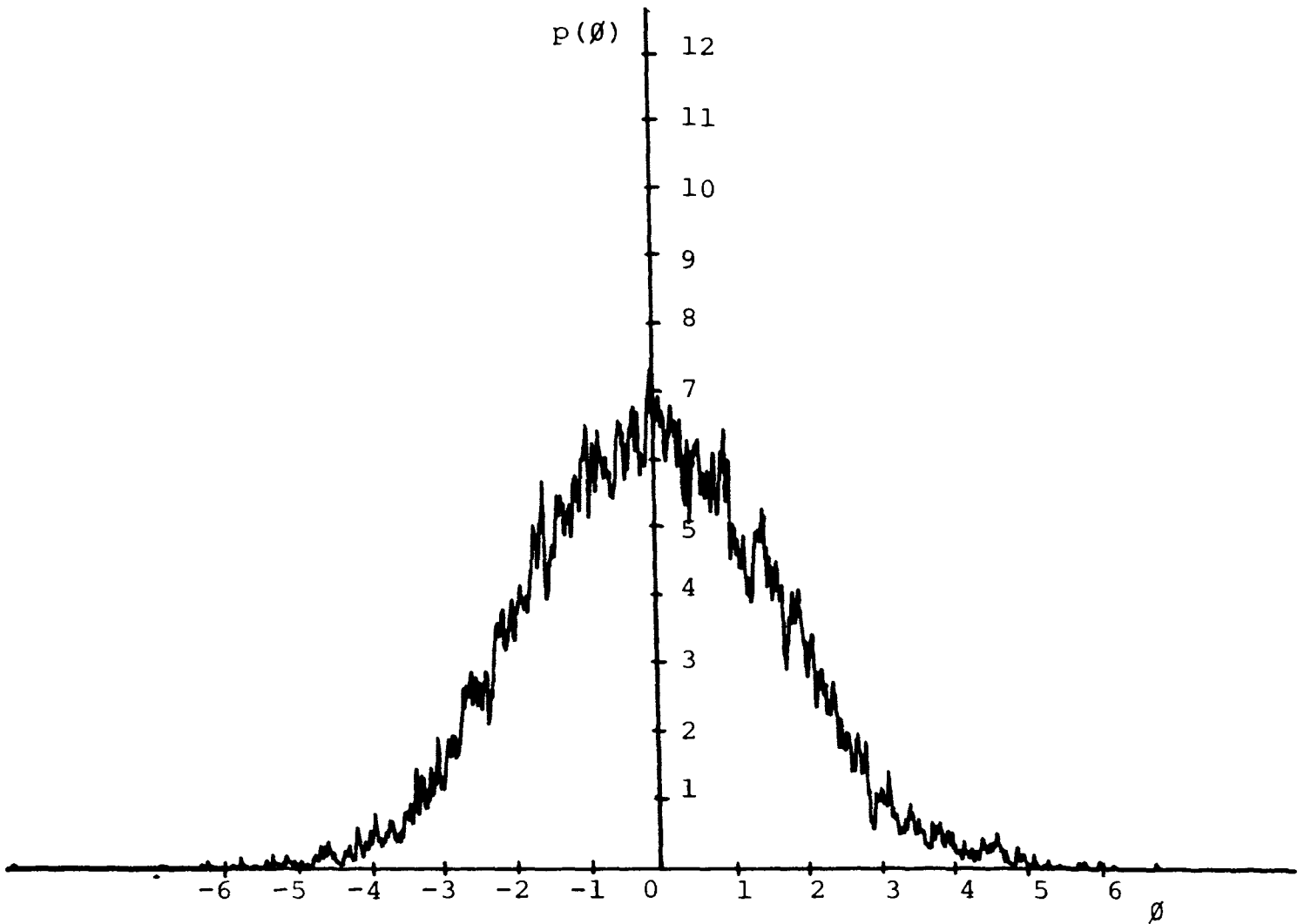
Figure 4. Probability Density Function of Phase Error;  $B_L=85\text{Hz}$ ;  $\rho=0.3$ ;  $m=1/2$ ,  $\text{SNR}=25\text{dB}$



Calibration: Abscissa; 1 unit = 1.12 radians  
Ordinate: 1 unit = .0431  $\text{rad}^{-1}$

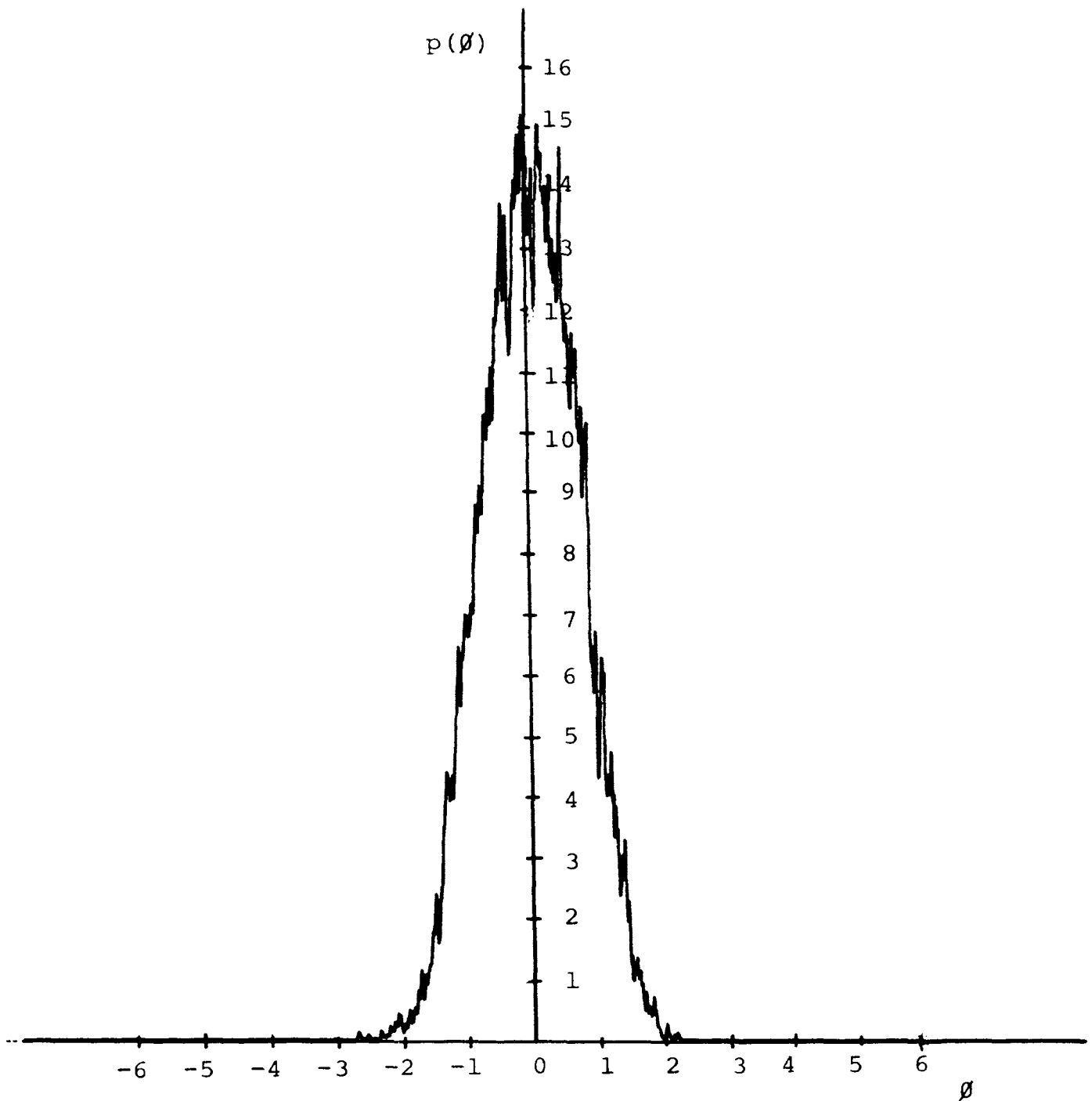
Figure 5. Probability Density Function of Phase Error;  
 $B_L=85\text{Hz}$ ;  $\rho=0.3$ ,  $m=1/2$ ,  $\text{SNR}=19\text{dB}$





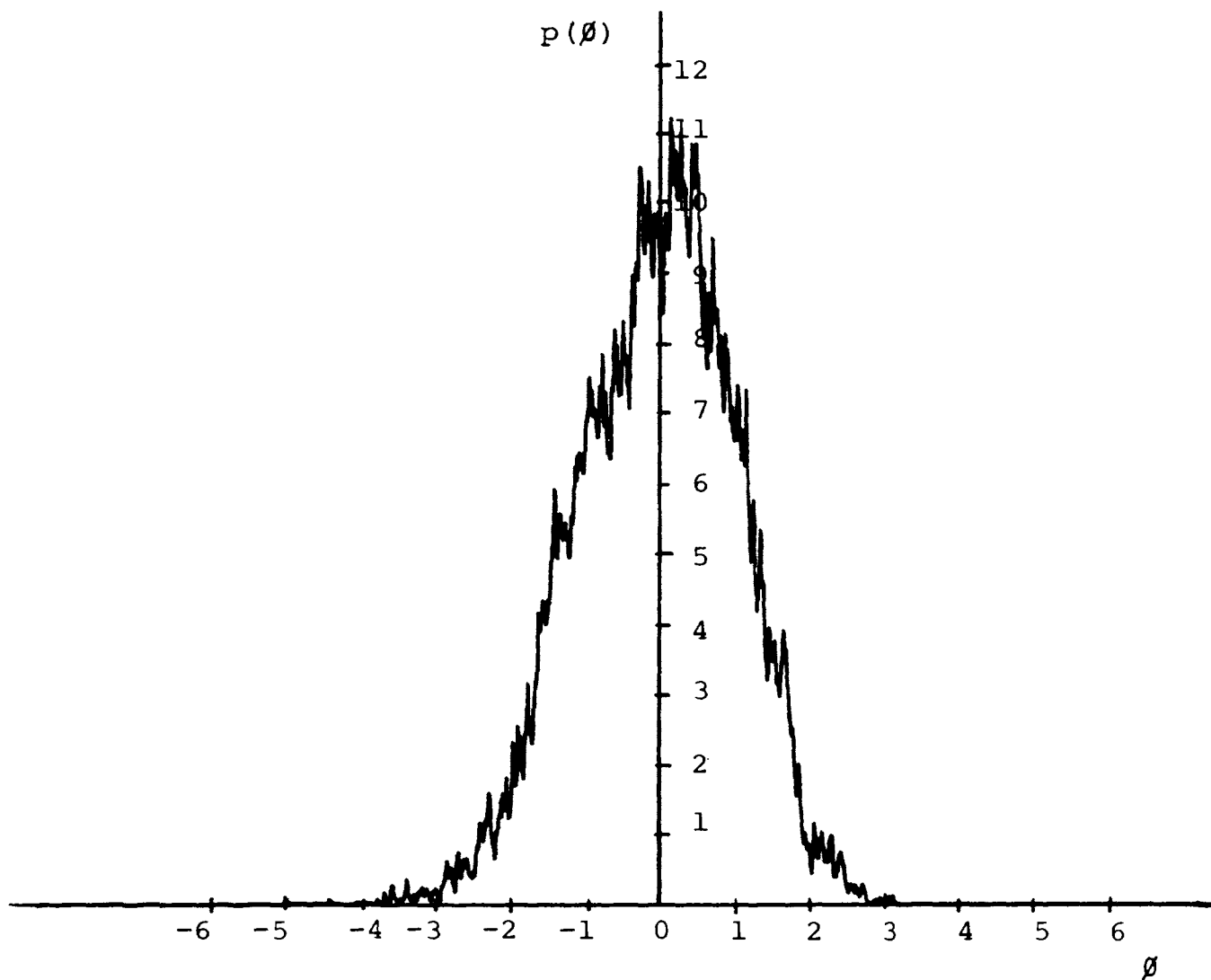
Calibration: Abscissa; 1 unit = 1.12 radians  
Ordinate: 1 unit = .0431 rad<sup>-1</sup>

Figure 6. Probability Density Function of Phase Error;  $B_L=85\text{Hz}$ ;  $\rho=0.3$ ,  $m=1/2$ ,  $\text{SNR}=17\text{dB}$



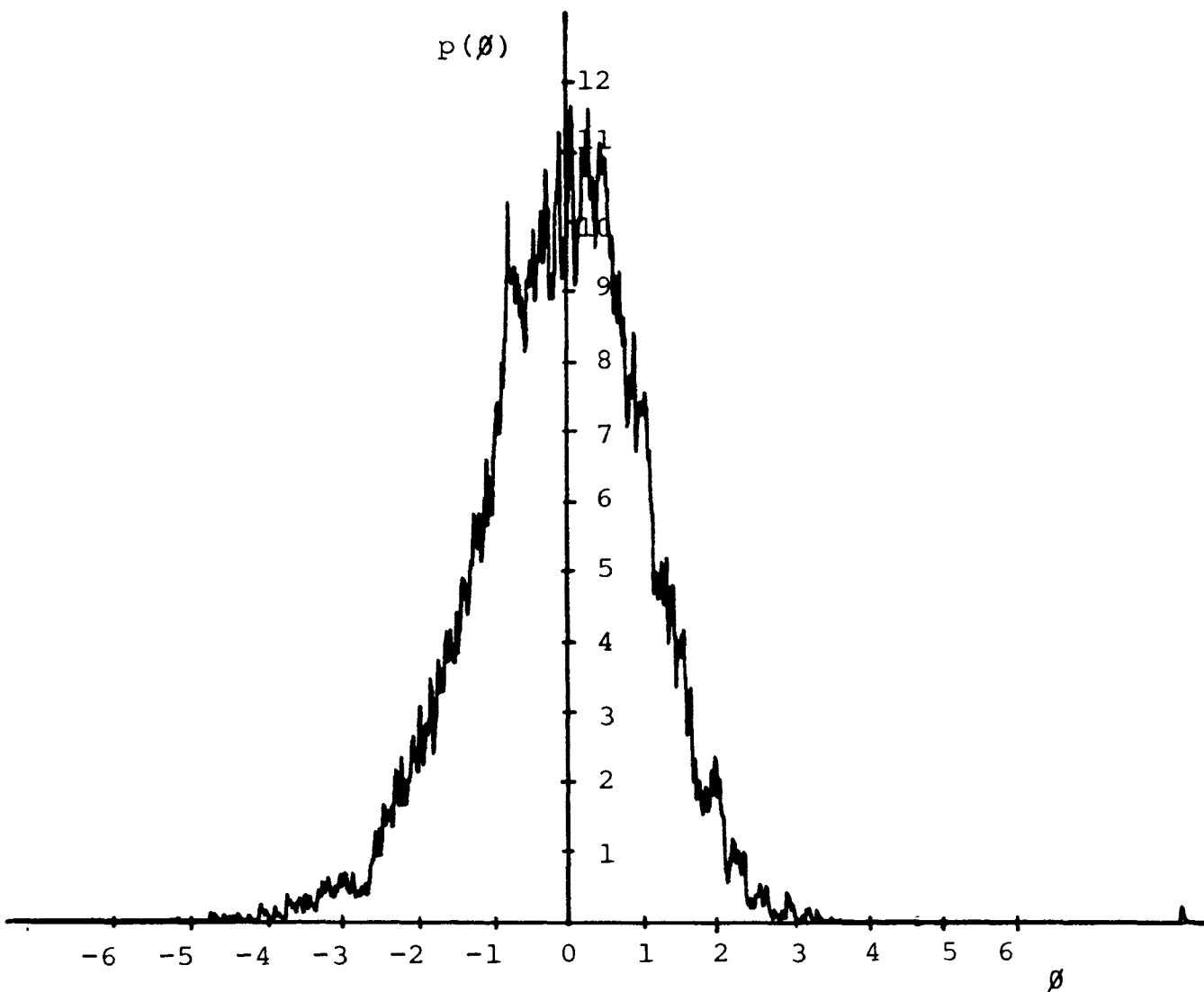
Calibration: Abscissa: 1 unit = 1.12 radians  
Ordinate: 1 unit = .031 rad<sup>-1</sup>

Figure 7. Probability Density Function of Phase Error;  
 $B_L=85\text{Hz}$ ;  $\rho=0.3$ ,  $m=1$ ,  $\text{SNR}=25\text{dB}$



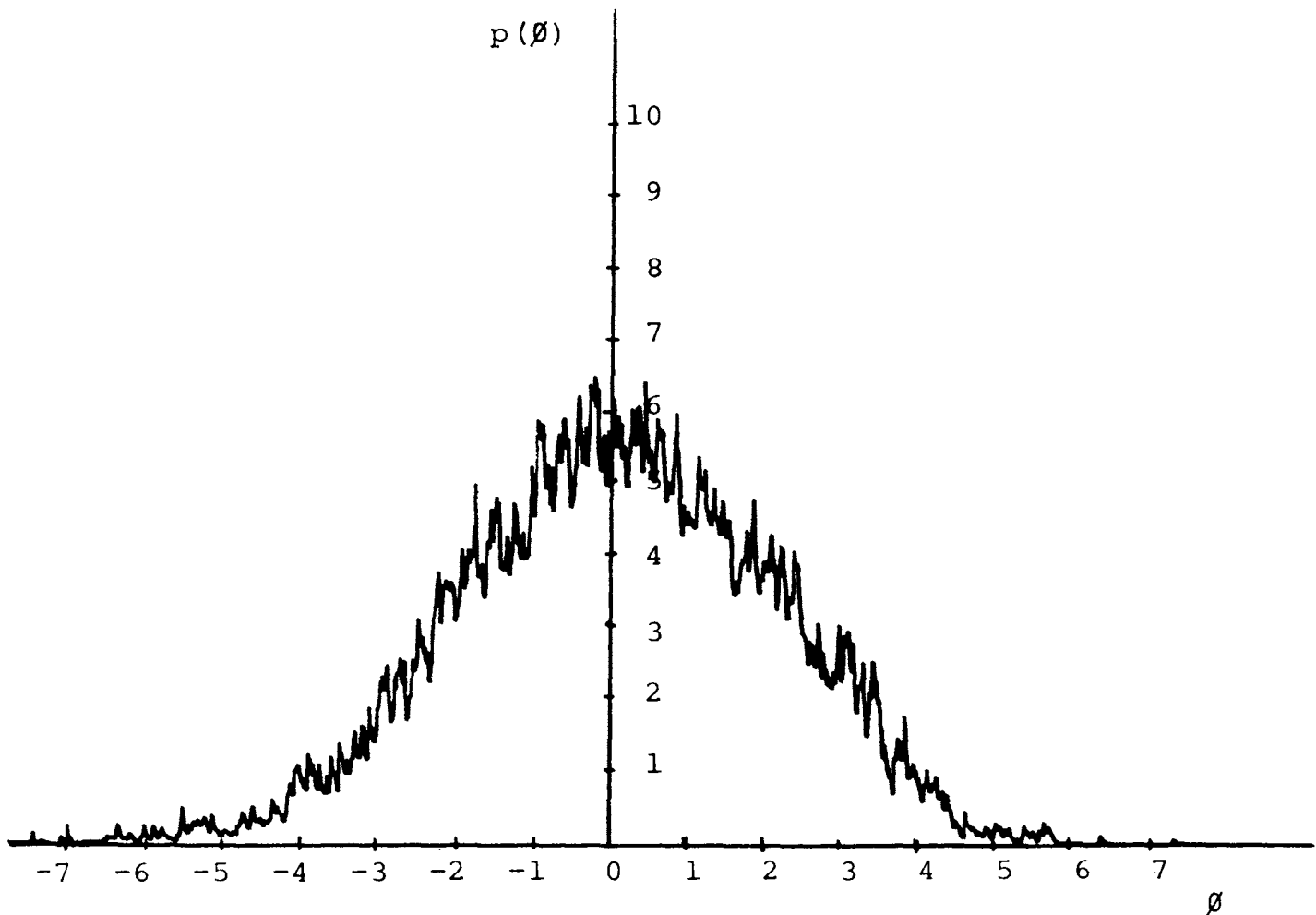
Calibration: Abscissa: 1 unit = 1.12 radians  
Ordinate: 1 unit = .0548  $\text{rad}^{-1}$

Figure 8. Probability Density Function of Phase Error;  $B_L=85\text{Hz}$ ;  $\rho=0.3$ ,  $m=1$ ,  $\text{SNR}=19\text{dB}$



Calibration: Abscissa: 1 unit = 1.12 radians  
 Ordinate: 1 unit =  $.0375 \text{ rad}^{-1}$

Figure 9: Probability Density Function of Phase Error;  $B_L=85\text{Hz}$ ;  $\rho=0.3$ ,  $m=1$ ,  $\text{SNR}=19\text{dB}$



Calibration: Abscissa: 1 unit = 1.12 radians  
Ordinate: 1 unit = .0376  $\text{rad}^{-1}$

Figure 10. Probability Density Function of Phase Error;  $B_L=85\text{Hz}$ ,  $\rho=0.3$ ,  $m=0.2$ ,  $\text{SNR}=25\text{dB}$

where

$$z = \frac{A^2}{N_o B_L} = \text{SNR}, \quad (27)$$

$$\beta = B_L / B_i ,$$

$N_o$  = Single sided power spectral density of the input noise,

and

$B_i$  = Single-sided low pass bandwidth of the input noise (essentially the bandwidth of the inphase and quadrature phase channel filters in Fig. 2)

Eq. (26) is obtained under the assumption that the phase error  $\theta$  is very small; thus, the loop can be linearized for the analysis.

Although there is significant difference between the values of the last two columns of Table I, we can make some observations in regard to the trend of the loop behavior when operated in the Costas, phase-lock or hybrid loop mode. When the modulation index is 1, the transmitted power is totally in the carrier. The loop operation is equivalent to a second-order phase-lock loop. Similarly, when the modulation index is 0.2, the transmitted power in the carrier is small and the loop operation is nearer to that of a Costas loop. When the modulation index is 0.5, the transmitted power is

TABLE I  
 TABULATION OF LOOP PHASE ERROR UNDER  
 VARIOUS CONDITIONS

m	SNR, dB	$\sigma_{\phi}$ , rad	$\sigma_H$ , rad
1	25	0.89	.055
	19	0.716	.127
	17	0.97	.14
0.5	25	1.13	.075
	19	1.20	.159
	17	1.36	.224
0.2	25	1.77	.1

Summary of parameters and variables:

m = modulation index

$\rho = 0.3$  (defined by (25))

SNR = Signal-to-noise ratio, noise measured in a loop bandwidth (defined by (27))

$\sigma_{\phi}$  = Experimentally measured phase error standard deviation

$\sigma_H$  = Theoretical phase error standard deviation (defined by (27))

distributed in the ratio of 1 to 3 between carrier and modulation components, and the loop is operating in a hybrid loop mode. When  $\beta$  is very small (of the order of  $10^{-3}$ ) and the SNR is high, the phase error standard deviation is maximum when the loop is operated in Costas loop mode, which was the case in the experiment. Under the same conditions of  $\beta$  and SNR, the phase error standard deviation of the hybrid loop has an intermediate value between that of the Costas and phase-lock loop.

The disagreement in the values of the last two columns of Table I can be attributed to the inaccuracy of the hybrid loop model employed and to experimental measurement error. In regard to model inaccuracy, Stewart assumed a first-order loop to derive the expression for  $\sigma_H^2$  given in (27). Charles and Lindsey have shown that the analysis for first order loop will also hold good for a second order loop without much loss of accuracy. The jitter from various sources contributes to the phase error in addition to the noise added to the signal and was not included by Stewart in his analysis. These additional jitter sources consist of the source jitter, the VCO jitter and finally the Exclusive-OR phase detector jitter. The phase jitter in the input signal may be due to the source itself or may be introduced during the frequency translation process. Although the VCO phase jitter was minimized by using a regulated power supply, it was not



completely eliminated. The exclusive-OR phase detector jitter might be due to the nature of its two inputs.

The measurement errors include such errors as calibration errors of the phase detector and probability density analyzer, inaccuracy in determining signal and noise levels at the loop input, and inaccuracies in loop parameters.

Another factor influencing the error was the type of noise generator used. In the calculation it was assumed that the added noise to the input signal was white. This was not true in the experiment. The power spectral density of the additive noise was not flat beyond 600-700 kHz. Also, the bandpass filter following the noise generator did not have an ideal flat response, as was assumed in the calculation.

It is interesting to compare the ratios of the experimental and computed phase-error standard deviations at the lowest SNR employed (17dB). At this SNR it is probable that most of the phase error was due to the added noise. At an SNR of 17dB, these ratios are,

$$\frac{\sigma_H |_{m=1}}{\sigma_H |_{m=0.5}} = 0.625 \quad \text{and} \quad \frac{\sigma_\emptyset |_{m=1}}{\sigma_\emptyset |_{m=0.5}} = 0.715$$

respectively. This is a relative comparison of the theoretical and experimental phase error standard deviations rather than the absolute comparison given in Table I. It

is seen that the results compare more favorably on this basis.

## CONCLUSIONS

An experimental model of hybrid phase-lock loop was built and was tested experimentally. There is disagreement between the theoretically computed values of phase error standard deviation and the values obtained experimentally. The difference in the values of the phase error standard deviations can be attributed to various errors of the experiment and also to the assumptions in the analysis. Attempts are being made to improve the hybrid loop built and to minimize the measurement errors.

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## VITA

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## APPENDIX A

## EXPERIMENTAL HYBRID LOOP COMPONENTS

a. Voltage Controlled Oscillator and  $90^\circ$  Phase Shifter

The circuit diagram for this module is shown in Fig. 11. In order to minimize the dependence of VCO frequency upon power supply voltage a one stage positive voltage regulator is used. The negative feedback principle is used; namely, an error voltage generated by zener comparison of the output of the regulator is amplified and fed back as a correction signal. A  $100\mu\text{fd}$  capacitor is used at the output of the regulator to reduce high frequency noise and transients. The zener used in the regulator has a value of 6.2 volts.

The voltage controlled oscillator consists of an astable multivibrator whose frequency is controlled by changing the charging current on the two cross-coupling capacitors by the input voltage. Potentiometers  $p_1$  and  $p_2$  are used to control the sensitivity of the VCO (voltage to frequency conversion rate) and the waveform symmetry. The VCO collectors are clamped to the 5 volts zener level by a pair of diodes to improve the frequency stability by establishing a firm logic level.

The outputs from the two collectors of the astable multivibrator are fed to two bistable multivibrators.

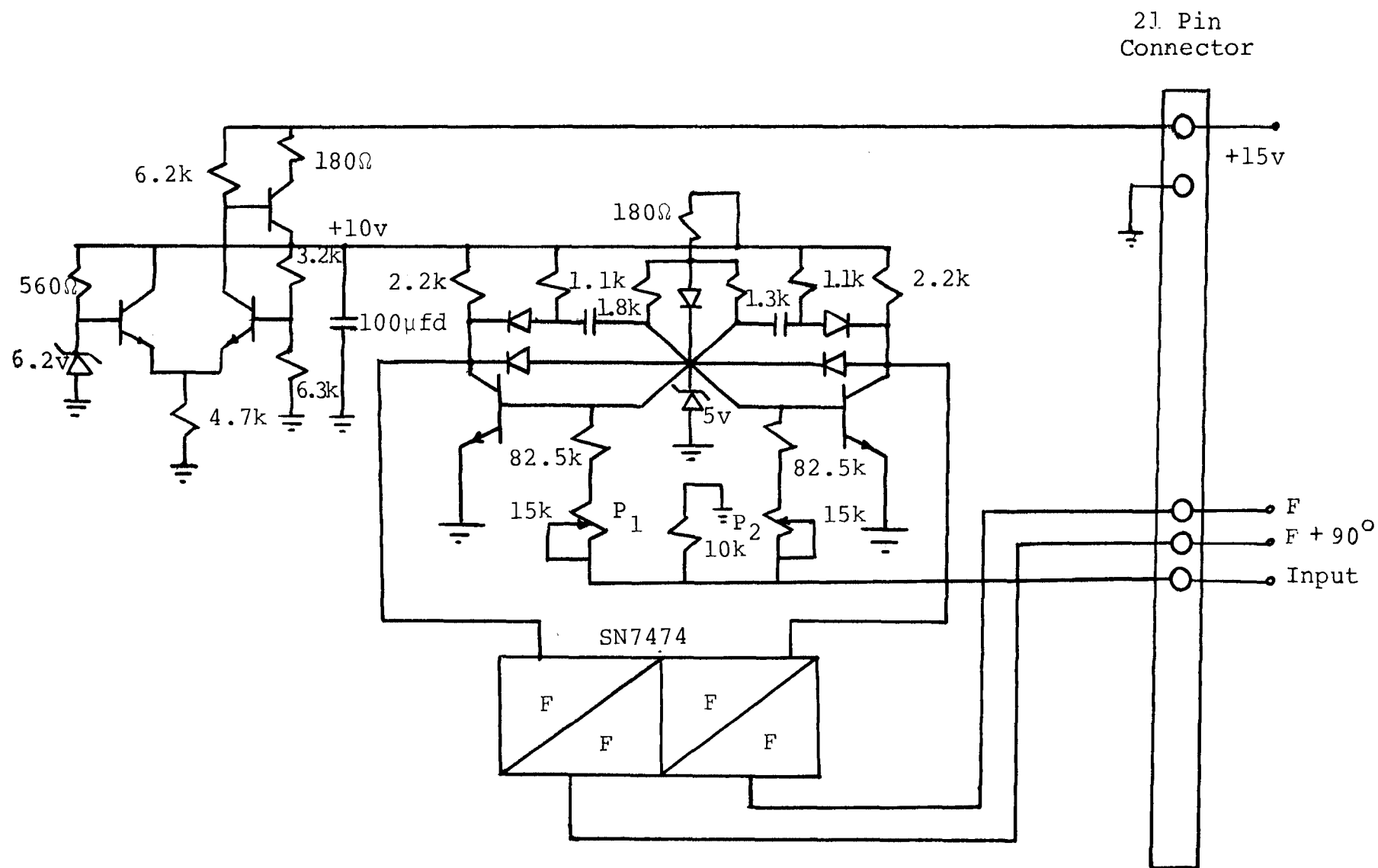


Figure 11. Voltage Controlled Oscillator



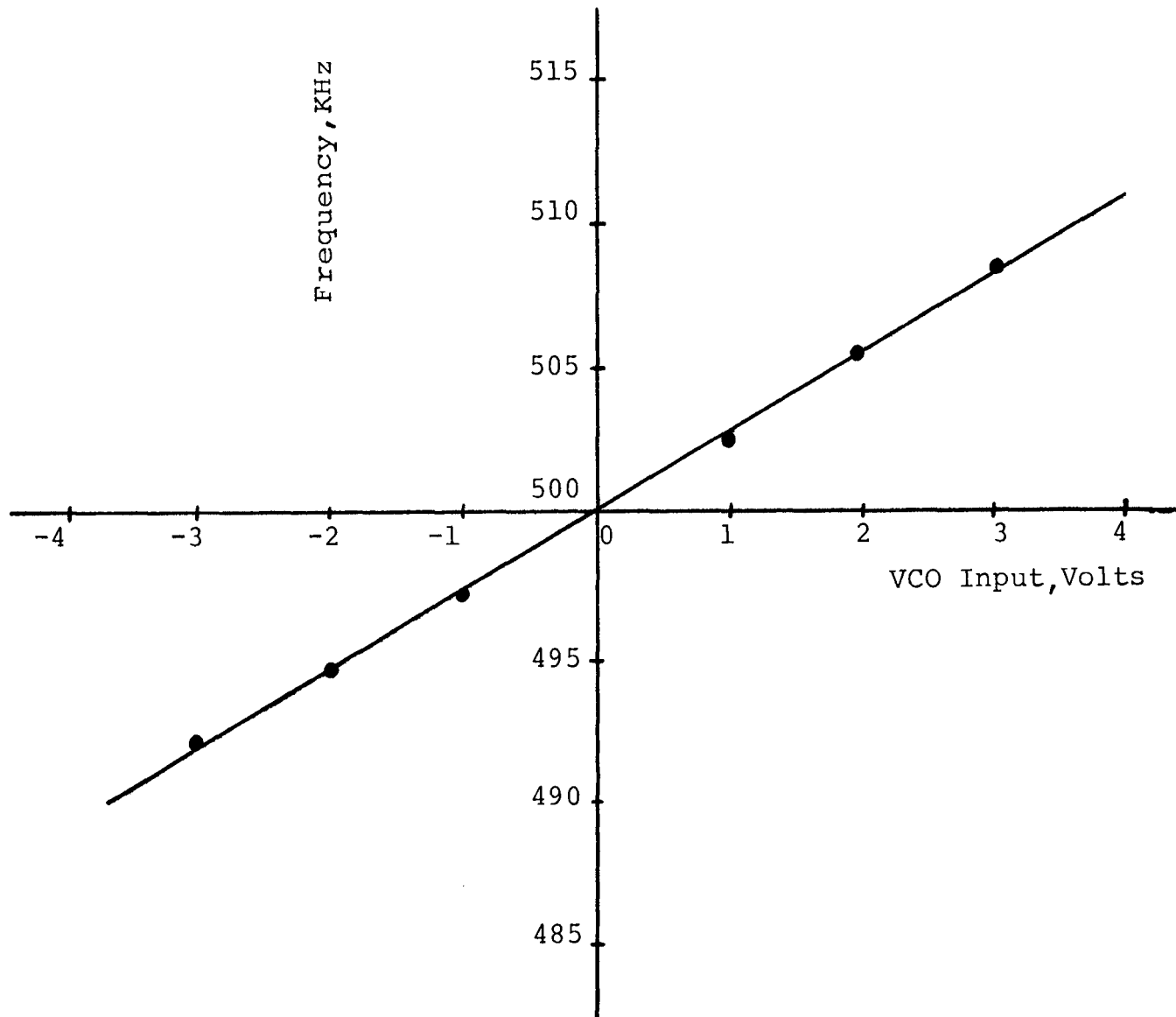


Figure 12. VCO Characteristic

Here IC chip-SN7474 which has two flip-flops triggered by positive going edges of the input waveform is used. As the inputs to the two flip-flops are  $180^\circ$  out of phase and as the flip-flops are triggered by the positive rising edge of the inputs, phase quadrature between the outputs of the two flip-flops is established. These signals are used as the reference inputs to the phase detectors of the hybrid loop.

b. Phase Detector

Multipliers M1 and M2 (Intronics Multipliers, Type M410) together with the outerloop filters  $G(p)$  form the phase detectors for the Costas loop and M1 in conjunction with  $F(p)$  form the phase detector for phase-lock loop (see Figures 13 and 14 for wiring diagrams). Filters  $G(p)$  are band pass filters with upper cut-off frequency at 16 kHz and lower cut-off frequency at 0.4 Hz. These phase detectors have a linear characteristic from zero to  $180^\circ$  ( $90^\circ \pm 90^\circ$ ) and have a gain constant of 1/10.

c. Adder, Loop Filter and Inverter

Three Intrinsic (Type A101) operational amplifiers (see Fig. 15) are used for adder, loop filter and inverter. Band switches  $B_1$  and  $B_2$  are provided to select loop bandwidth of 8.5Hz, 85Hz, 850Hz, and 8.5kHz. Potentiometers  $P_1$  and  $P_2$  are provided to adjust the phase-lock

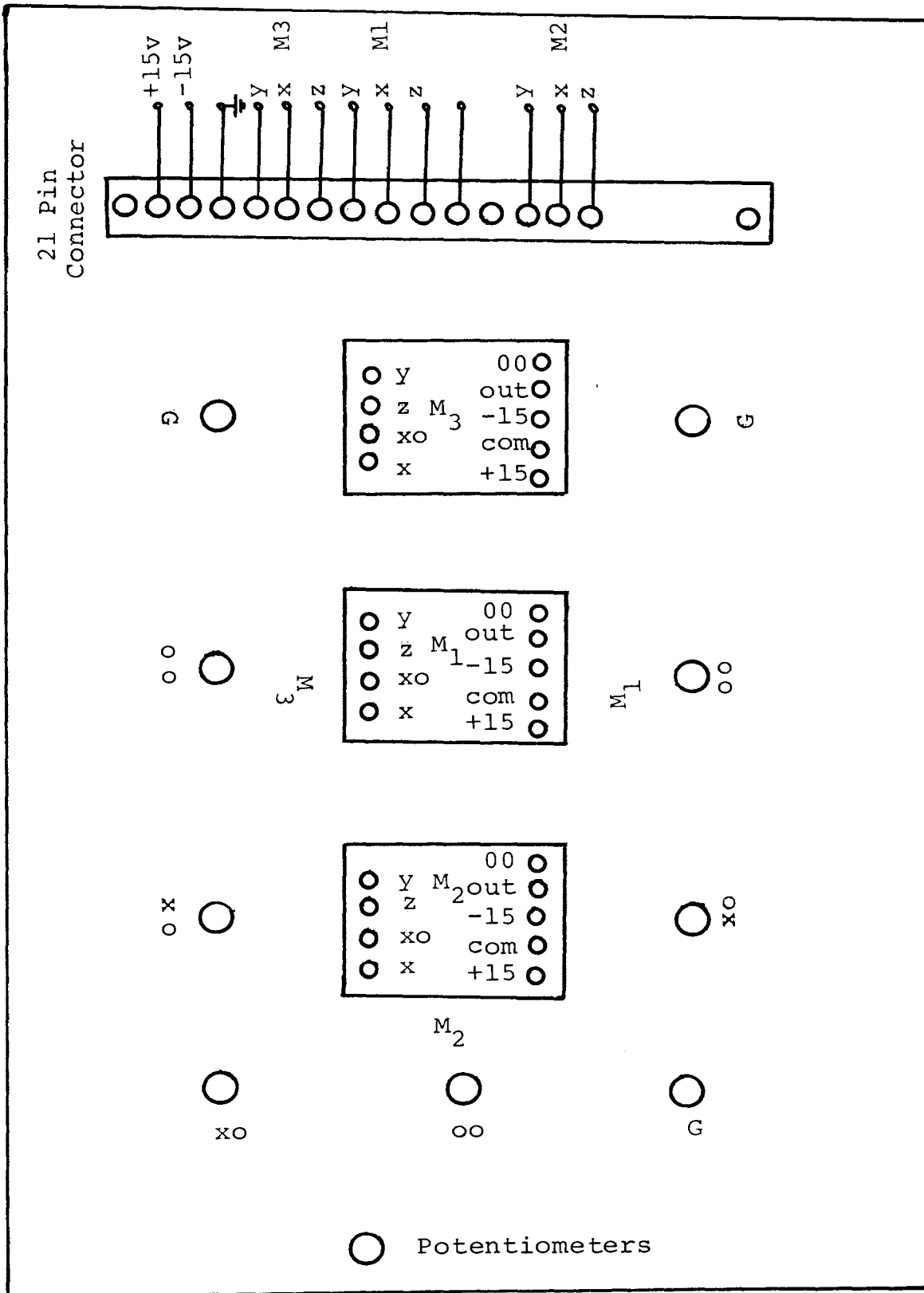


Figure 13. Phase Detectors

Numbers in parentheses denote pin numbers not shown on connector diagram

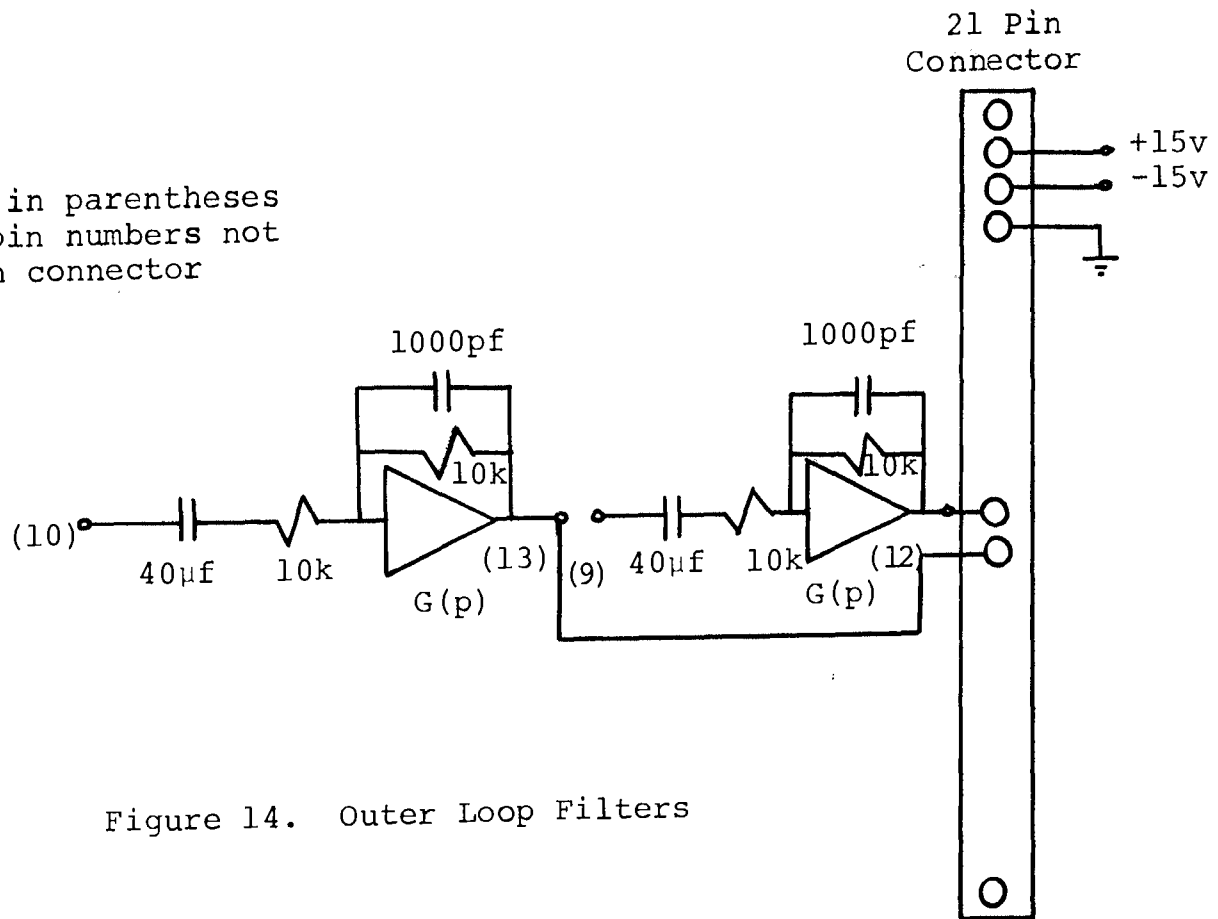


Figure 14. Outer Loop Filters

Numbers in parentheses  
denote pin numbers not  
shown on connector diagram

21 Pin  
Connector

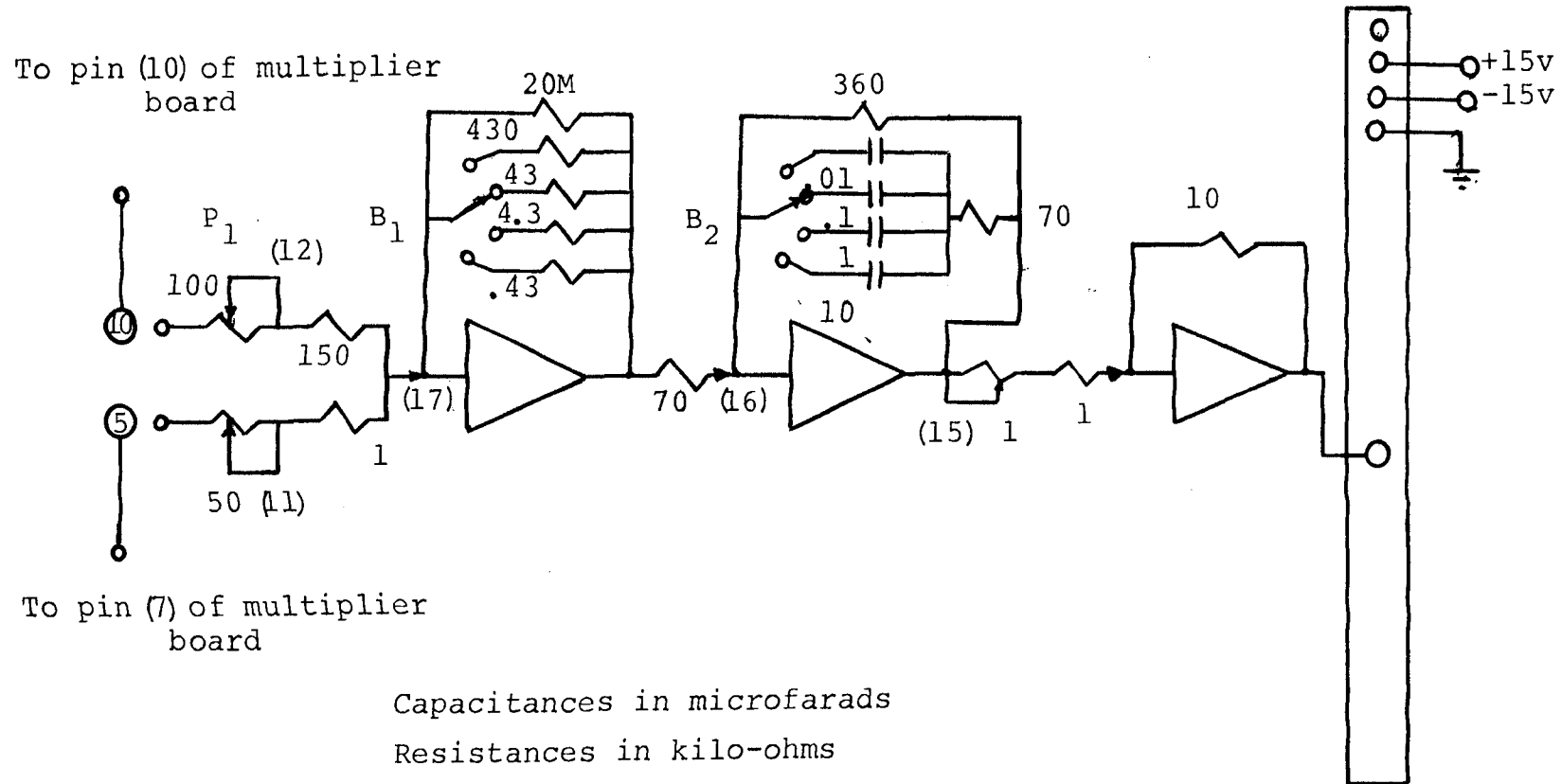


Figure 15. Adder, Loop Filter and Inverter

loop gain  $a_1$  and Costas loop gain  $a_2$  to the required value. With different settings of these two potentiometers, we obtain different values of  $\rho$ , which is proportional to the ratio of  $a_2$  to  $a_1$ . The loop filter used here has a transfer function of the form

$$F(p) = \frac{p+a}{p+\epsilon}$$

where  $a$  is the filter time constant and  $\epsilon$  is the filter imperfection factor. In Table II the values of loop bandwidth, and damping factor  $\xi$  for various positions of  $B_1$ ,  $B_2$ ,  $P_1$  and  $P_2$  are shown.

APPENDIX B  
DESIGN EXAMPLE

A damping factor of 0.707 and loop band-width of 850Hz are chosen in the following design example.

(i) Let  $\epsilon/G_o = 0.1$

then,

$$B_L = \frac{G_o + a}{4(1 + \epsilon/G_o)}$$

$$= \frac{G_o + a}{4.4}$$

(ii) Calculation of  $a/G_o$

$$\sqrt{2a/G_o} = 1 + \epsilon/G_o = 1.1$$

Therefore,  $a/G_o = 0.605$

(iii) Calculation of  $G_o$

$$G_o = \frac{4\sqrt{2a/G_o}}{1 + a/G_o} B_L$$

$$= \frac{4 \times 1.1}{1.605}$$

$$= 2740$$

(iv) Determination of loop filter time constants.

$$a = 0.605 G_O = 1660$$

$$\epsilon = 0.1 G_O = 274$$

$$\text{Now, } F(p) = K_3 \frac{p+a}{p+\epsilon}$$

$$\text{where, } K_3 = R_3/R_1 \cdot R_2/R_2+R_3$$

$$a = 1/R_2 C \quad \text{and} \quad \epsilon = \frac{1}{(R_2+R_3)C} \cdot$$

$$\text{Let } C = 0.01 \mu\text{fd}$$

$$\text{Then, } R_2 = 1/aC$$

$$= \frac{10^7}{1660} = 60\text{k}$$

$$R_2 + R_3 = \frac{10^7}{274}$$

$$= 365\text{k}$$

$$R_3 = 305\text{k}$$

$$R_1 = 60\text{k}$$

so that  $k_3$  is nearly unity.

(v) Computation of gains  $a_1$  and  $a_2$

$$G_O = k_v k_3 a_1 k_o k_1^A [m + \rho(1-m^2)]$$

$$\text{Let } m = 1/2$$

$$\rho = 0.66$$

$$k_o = 1/10$$

(approximately for optimum operation as obtained from Stewart [5])



$$k_1 A = 3$$

$$m + \rho(1 - m^2) = 1$$

(approximate relation  
for optimized operation)

Therefore,

$$a_1 = \frac{2740 \times 0.83}{2.2 \times 10^4 \times 1/10 \times 3}$$

$$= 0.357$$

$$a_2 = \frac{0.66 \times 0.357}{1/100 \times 3}$$

$$= 7.6$$

Similarly the values for other loop bandwidths are calculated and they are tabulated in Table 2.

TABLE II  
SUMMARY OF LOOP COMPONENT VALUES

$\epsilon$	C	a	Total Gain	R	$a_1$	$a_2$	$G_o$	$B_L$	$\xi$	
2300	0.001 $\mu$ F	14300	43	450	MAX	.003	.45	23.2	35Hz	2
					MIN	.0018	.009	14.0	21.6Hz	2.6
				4.5k	MAX	.03	4.5	232	330 <sup>1</sup> Hz	0.695
					MIN	.018	.09	140	310Hz	0.86
				45k	MAX	.3	4	2320	2.08kHz	0.4
					MIN	.18	.9	1400	1.58kHz	0.415
				450k	MAX	3	450	23200	8.5kHz	.702
					MIN	1.8	9	14000	6.1kHz	.575
230	0.01 $\mu$ F	1430	4.3	450	MAX	.003	.45	2.32	3.5Hz	2
					MIN	.0018	.009	1.40	2.16Hz	2.6
				4.5k	MAX	.03	4.5	23.2	33Hz	0.695
					MIN	.018	.09	14.0	31Hz	0.86
				45k	MAX	0.3	45	232	208Hz	0.4
					MIN	0.18	.9	140	158Hz	0.415
				450k	MAX	3	450	2320	850Hz	.702
					MIN	1.8	9	1400	610Hz	.575
23	0.1 $\mu$ F	143	4.3	450	MAX	.003	.45	232		
					MIN	.0018	.009	1.4		
				4.5k	MAX	.03	4.5	23.2	8.5Hz	.4
					MIN	.018	.09	14.0	6.1Hz	.415
				45k	MAX	.3	45	232	85Hz	.702
					MIN	.18	.9	140	61Hz	.575
				450k	MAX	3	450	2320	616Hz	2.04
					MIN	1.8	9	1400	386Hz	1.58
2.3	1 $\mu$ F	14.3	4.3	450	MAX	.003	.45	2.32		
					MIN	.0018	.009	1.4		
				4.5k	MAX	.03	4.5	23.2	8.5Hz	.702
					MIN	.018	.09	14.0	6.1Hz	.575
				45k	MAX	.3	45	232	58Hz	2
					MIN	.18	0.9	140	35Hz	5
				450k	MAX	3	450	2320		
					MIN	1.8	9	1400		

$$0.05 \leq \rho \leq 2.5$$

APPENDIX C  
PHASE ERROR PROBABILITY DENSITY MEASUREMENT

A block schematic of the set up used to measure the phase error density is given in Figure 16. Noise from the noise generator is passed through the band pass filter with equivalent noise bandwidth  $B$ ; the output of the BPF is added to the received signal which then is the input to the hybrid loop under test. The unmodulated reference from the source is hard limited in two stages of amplification, and then used as a reference input to the phase detector. The voltage controlled oscillator output, whose phase is to be compared with that of the reference, is the second input to the phase detector.

The phase detector itself consists of two flip-flops and four nand gates. Flip-flops are used for further wave shaping of the VCO output and the reference signal as well as to double the range of the phase detector. The four nand gates are connected to form an exclusive-OR gate which functions as the phase detector. The squarer and the phase detector used are shown in Figure 17.

The high frequency components of the phase detector output are removed by a low-pass filter which is sufficiently wideband to pass the phase error undistorted and then given to the probability density analyzer. The output of the probability density analyzer is recorded with the help of

an XY Plotter and the resulting phase-error probability plots are shown in Figures 4-10. Figures 4-10 were obtained with the loop bandwidth set at 85Hz, input signal level at 0.6v rms, PLL gain  $a_1$  at 0.3, Costas loop gain  $a_2$  at 11 (resulting  $\rho$  is 0.3) and the total gain at 4.3. The corresponding settings with respect to Fig. 2 were, band-switches  $B_1$  and  $B_2$  at position 2, potentiometer  $P_1$  at 0, potentiometer  $P_2$  at 3.10 and  $P_3$  at 9.50.

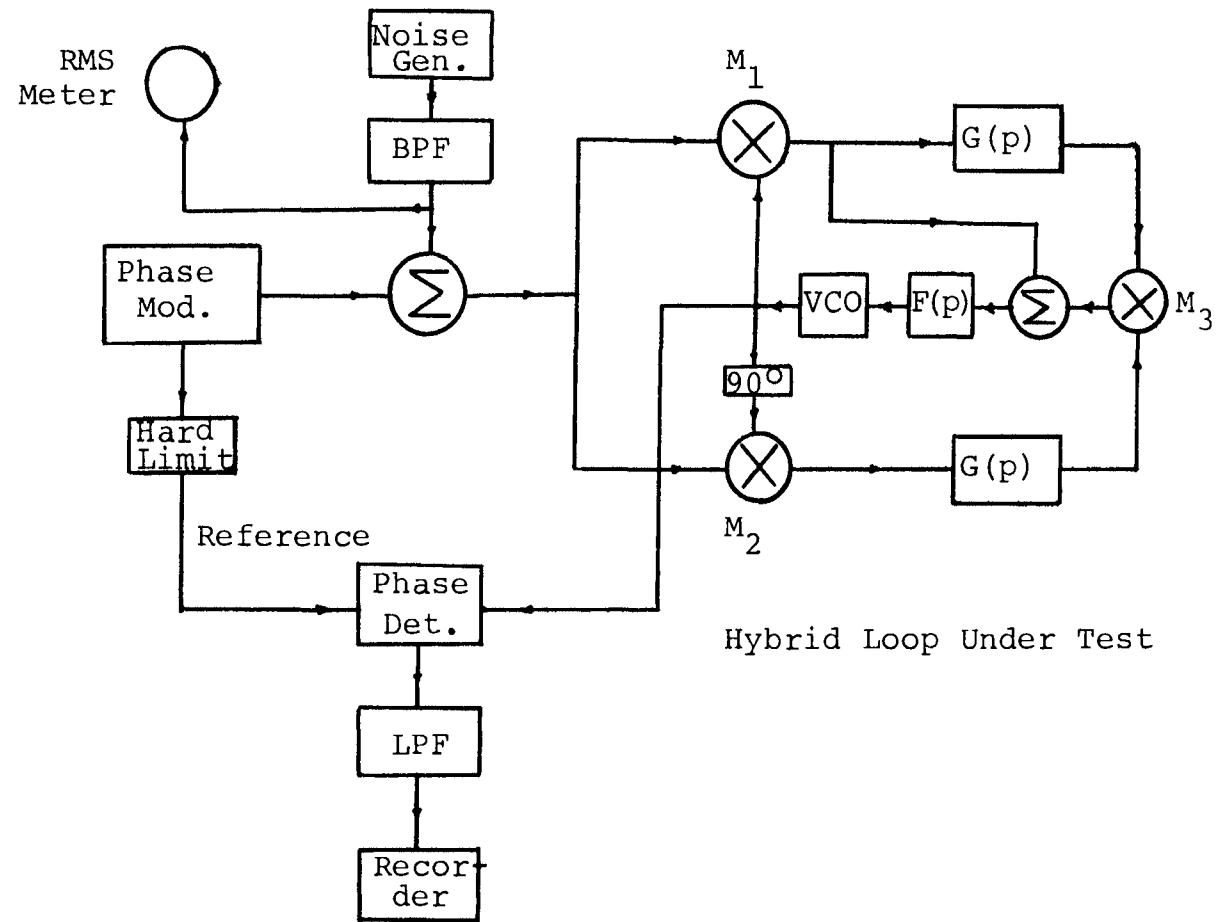


Figure 16. Experimental Set Up for Measurement of Phase Error Probability Density

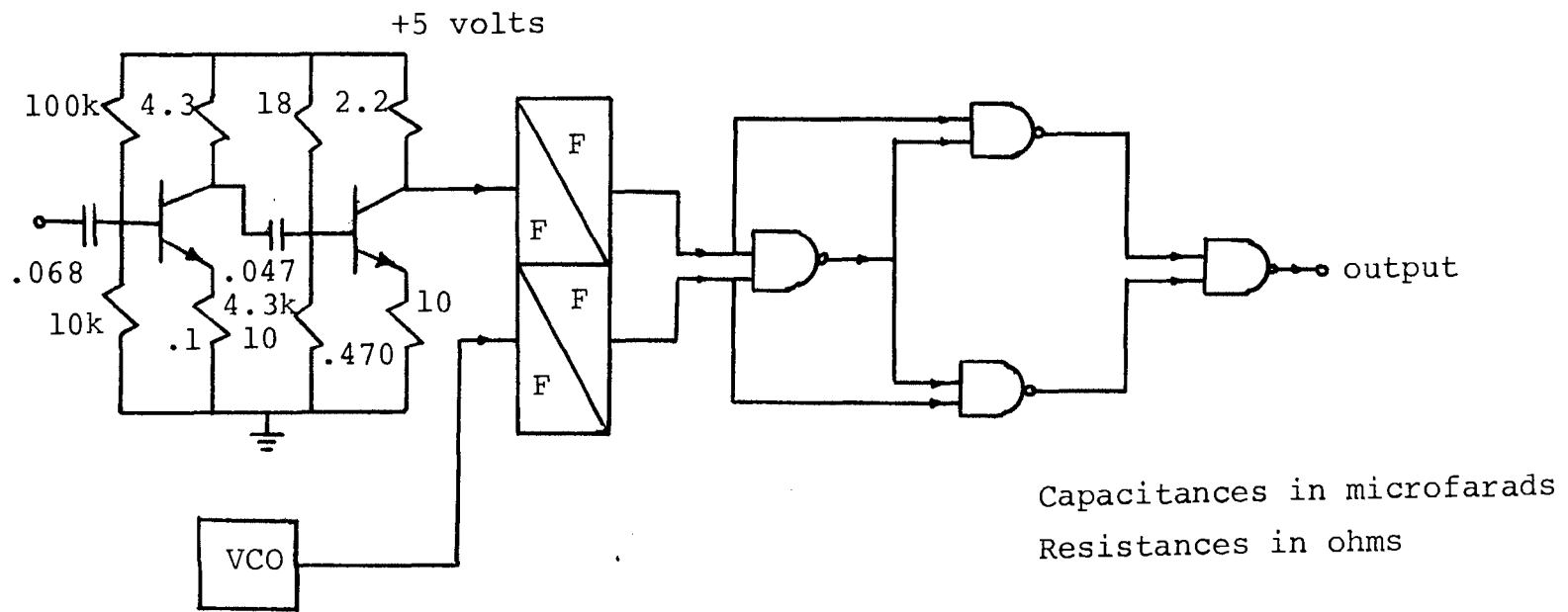


Figure 17. Exclusive-OR Phase Detector

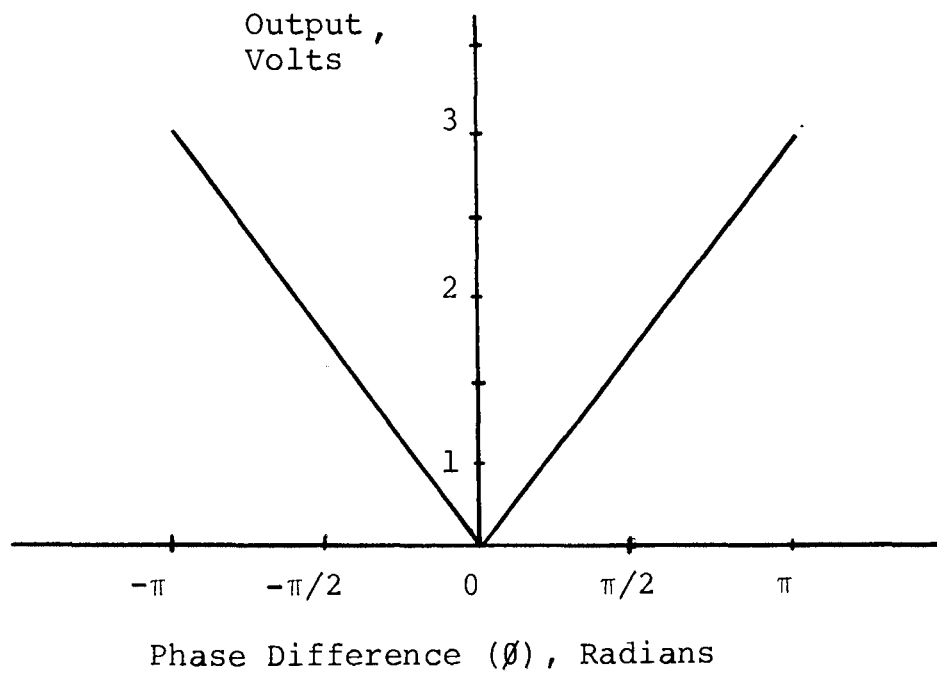


Figure 18. Exclusive-OR Phase Detector Characteristic