

Implementation of an FSK Modem Using the TMS320C17

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Abstract

This report presents a complete hardware design for a splitband modem, and the software to implement a V.21/Bell 103 300-bps modem, using a TMS320C17 DSP.

- ❑ The first section reviews basic modem concepts and definitions and introduces frequency shift keying (FSK) data modulation.
- ❑ The second section describes the major functional blocks of the FSK modem system.
 - Host Interface
 - Modem Controller
 - Digital Signal Processor
 - Analog Front End
- ❑ The third section discusses DSP software implementation of the V.21/Bell 103 300-bps modem, using a TMS320C17 DSP.
- ❑ The fourth section reviews some issues involved with incorporating additional code into DSP software provided in Appendix B.
- ❑ The fifth section summarizes conclusions.
- ❑ Appendix A is a derivation of the filter coefficient value required for the sample fraction time delay.
- ❑ Appendix B is the source code listing for the TMS320C17 modem implementation.

The report also includes flow chart and tabular frequency and phase step data.



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Introduction

This application report presents an implementation of a 300-bit-per-second (BPS) modem conforming to the V.21 and Bell 103 standards, using a TMS320C17 Digital Signal Processor (DSP).

The purpose of this application report is, with references [1], [2], [3], to provide a complete hardware design for a splitband modem and the software to implement a V.21/Bell 103 300-bps modem. The designer can then concentrate on developing value-added functions, such as V.22bis or V.22 standard modems, encryption algorithms, etc. These value-added functions are implemented in software and can be easily incorporated into the TMS320C17 software provided in Appendix B.

The structure of this report is as follows:

- The first section reviews basic modem concepts and definitions and introduces the reader to frequency shift keying (FSK) data modulation.
- The second section describes the major functional blocks of the FSK modem system presented in this report:
 - Host interface,
 - Modem controller,
 - Digital signal processor, and
 - Analog front end.
- References to documents describing the actual hardware implementation are provided.
- The third section discusses the DSP software implementation of the V.21/Bell 103 modulator/demodulator using the TMS320C17 DSP.
- The fourth section reviews some of the issues involved with incorporating additional code into DSP software provided in Appendix B.
- The fifth section concludes this report.
- Appendix A is a derivation of the filter coefficient value required for the sample fraction time delay.
- Appendix B is the source code listing for the TMS320C17 modulator and demodulator implementation.

Background

Over the past decade there has been a proliferation in the number and the use of computer systems. Accompanying this growth, there has been an increased demand for data communications between the various computer systems and terminals.

One of the most convenient and frequently used methods of data communications between geographically separated computer equipment is via the Public Switched Telephone Network (PTSN). The essential element for this method of data communication is the modem.

The modem converts the digital data it receives from the computer system or terminal into a modulated analog signal that is transmitted via the telephone network to the destination computer system or terminal. At the destination, the receive modem demodulates the received signal and transfers the digital data to the receiving terminal or computer system.

Table 1 shows a number of popular modem standards as specified by either the International Telegraph and Telephone Consultative Committee (CCITT) or the Bell System.

Table 1. Bell and CCITT Modem Standards

Modem	Standard	Type*	Modulation	Data Rate (BPS)	Duplex
Bell	103	S/B	FSK	300	Full
	202	S/B	FSK	1200	Half
	212A	S/B	DPSK	1200	Full
	201	S/B	DPSK	2400	Half
CCITT	V.21	S/B	FSK	300	Full
	V.22	S/B	DPSK	1200	Full
	V.22bis	S/B	QAM	2400	Full
	V.32	E/C	QAM	9600	Full

* S/B = Split band E/C = Echo Cancelling

Modems can be either half-duplex or full-duplex. In a half-duplex system, the transmission can be in either direction; however, only one direction is possible at a time. A half-duplex modem cannot simultaneously transmit and receive information. At the end of its transmission sequence, the modem must advise the receiving modem that the sequence is complete. The receiving modem may then begin transmitting data.

In a full-duplex system, the data transmission is bidirectional. Both modems may simultaneously transmit and receive data. Bidirectional (simultaneous data transmission) is achieved by either splitband or echo cancellation techniques.

Figure 1 shows the spectral response of a typical telephone channel. A splitband modem uses a filtering scheme to separate the telephone channel into two distinct frequency bands. One band is dedicated to the transmissions of the originate modem, the other band is dedicated to transmissions of the answer modem. To separate the received signal from the received and transmitted signal that is detected on the two-wire telephone line, the modem removes the transmitted signal frequency band using a splitband filter [1], [4], or by other means (such as software implemented on the DSP). Dividing the telephone channel into two separate non-overlapping frequency bands limits the maximum baud rate.

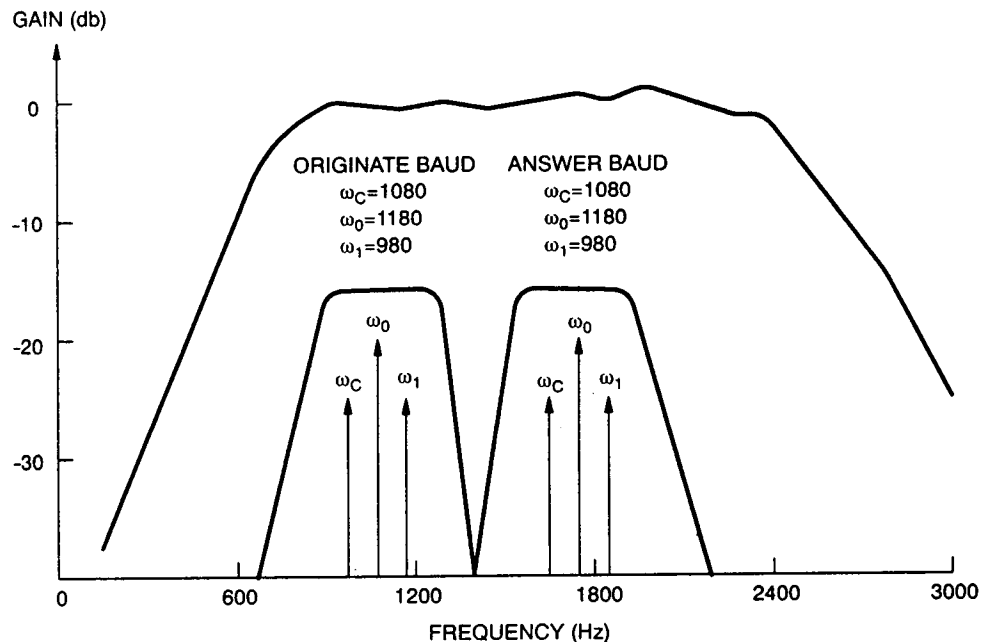


Figure 1. Spectral Response of a Typical Telephone and a V.21 Splitband Modem

The actual bit rate of the channel is determined by the baud rate and the data modulation scheme that is employed. Splitband type modems are typically used in low- to moderate-speed applications. As shown in Table 1, each modem standard uses a particular modulation scheme. For example, CCITT V.21, V.22, and V.22bis standards specify the frequency shift keyed (FSK), phase shift keyed (PSK) and quadrature amplitude modulation (QAM) schemes respectively.

Echo cancellation type modems, such as V.32, transmit both the originate and answer signals on the same channel. This allows both the originate and answer modems to utilize the complete bandwidth of the channel and to maximize the data baud rate. It is still necessary to separate the receive signal from the receive and transmit signal detected on the two-wire telephone line. However, the originate and answer signals are superimposed on the same channel band, and separating techniques that are more sophisticated than those found in splitband-type modems are required. The fact that transmit signal is typically 20 dB stronger than the receive signal, as measured on the transmit Tip and Ring, further complicates the extraction of the receive signal.

Echo cancellation type modems use algorithms that subtract an estimate of the transmit signal from the signal sampled from the two-wire telephone line, to determine the receive signal. Refer to [5] and [6] for further information on Echo cancellation type modems.

Table 2 shows the transmission frequencies for answer and originate modes for both the binary FSK modulated 300-bps V.21 and Bell 103 standards. It also shows details of the V.23 and Bell 202 1200-bps half-duplex standards.

Table 2. Binary FSK Transmission Frequencies

Modem Standard		Carrier (Hz)	1(Mark) (Hz)	0(Space) (Hz)
V.21	Originate	1080	980	1180
	Answer	1750	1650	1850
BELL 103	Originate	1170	1270	1070
	Answer	2125	2225	2025
V.23		1700	1300	2100
BELL 202		1700	1200	2200

Since this report is primarily concerned with the 300-bps V.21 and Bell 103 standard modems, it is worthwhile to review FSK data communication.

These are the primary advantages of an FSK system:

1. There is no requirement for carrier phase recovery; this reducing system complexity.
2. Increased immunity to amplitude nonlinearities. FSK is a constant envelope signal, with the information transmitted in the zero crossings. It is less affected by amplitude nonlinearities than amplitude modulated schemes, and
3. The modulator and demodulator architectures are easily implemented in software.

The primary disadvantage of FSK modulation is its low spectral efficiency. Because the telephone network is bandlimited to 4KHz, only moderate data transmission rates over the telephone network are supported by an FSK modulation scheme. As a consequence, FSK is often the favored modulation scheme for very low cost, low-to-moderate speed data communication systems.

Subsequent sections of this report discuss FSK modulation and demodulation in some detail. It is important that you understand the mathematical representations of FSK signals. FSK modulation is represented in the following manner:

$$S(t) = \cos((\omega_c \pm \delta\omega) * t + \phi) \quad (1)$$

where $S(t)$ = Transmitted signal
 ω_c = Carrier frequency
 $\delta\omega$ = Frequency shift
 t = Time
 ϕ = Phase shift

For a given baud period T , $S(t)$ is at a frequency $f_1 = (f_c + \delta f)$ or $f_0 = (f_c - \delta f)$, corresponding to the transmission of a 1 or 0, respectively, for the duration of the baud period. In some cases, it is convenient to represent

$$\begin{aligned} \omega_0 &= \omega_c - \delta\omega \\ \omega_1 &= \omega_c + \delta\omega \end{aligned} \quad (2)$$

Thus the following identities are true:

$$\begin{aligned} \omega_c &= (\omega_1 + \omega_0)/2 \\ \delta\omega &= (\omega_1 - \omega_0)/2 \end{aligned} \quad (3)$$

Some binary FSK modulation schemes, such as V.21, have ω_0 greater than ω_1 ; so by (3), $\delta\omega$ would be negative. Figure 2 shows an FSK signal transmission.

Note that the telephone channel provides limited spectral bandwidth. To achieve progressively higher data rates, more spectrally efficient modulation schemes, such as PSK and QAM, must be used. As spectral efficiency increases, typically, the complexity of the signal modulation and demodulation schemes increase. Additional information on modulation schemes can be found in references [4], [5], [6] and [7].

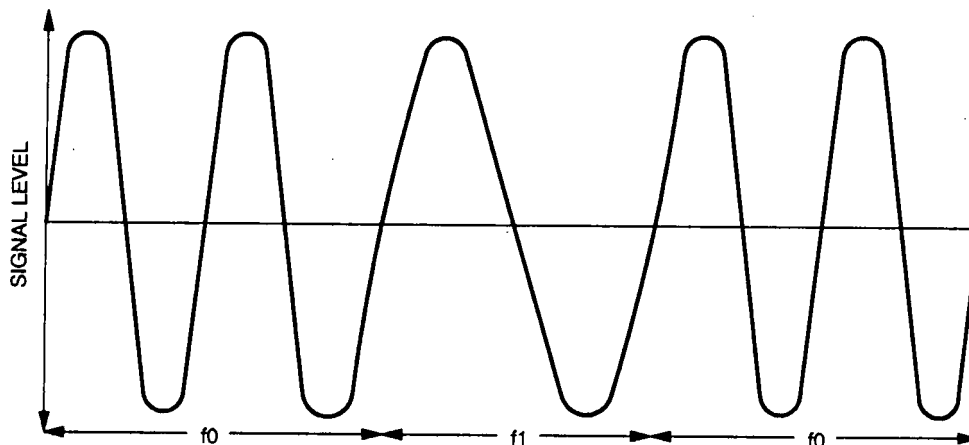


Figure 2. FSK Signal Transmission

System Description

As discussed in the introduction, this application report presents the implementation of a V.21/Bell 103 300-bps FSK modem using a TMS320C17 Digital Signal Processor. The system hardware is identical to that of the Texas Instruments DSP2400 modem [1].

There are significant functional differences between the modem design provided here and the DSP2400 modem. These result from the differences between the TMS320 code provided in Appendix B and the DSP2400 code. The software found in Appendix B implements a V.21/Bell 103 FSK modem. The DSP2400 also implements V.22, Bell 212A, and V.22bis standard modems that implement PSK and QAM modulation/demodulation and the associated carrier recovery, clock recovery, and adaptive equalization functions.

The software in Appendix B provides all the necessary hooks so that the designer can easily incorporate his own custom value-added features (such as V.22 and V.22bis standard modems). Nevertheless, the reader should be aware of the difference between the DSP2400 software implementation and the software in Appendix B, particularly when referring to any DSP2400 related literature [1], [2], [3].

Figure 3 is a block diagram showing the components of the modem system. The modem consists of the following subsystems:

1. Host interface
2. Modem controller
3. Digital signal processor
4. Analog front end

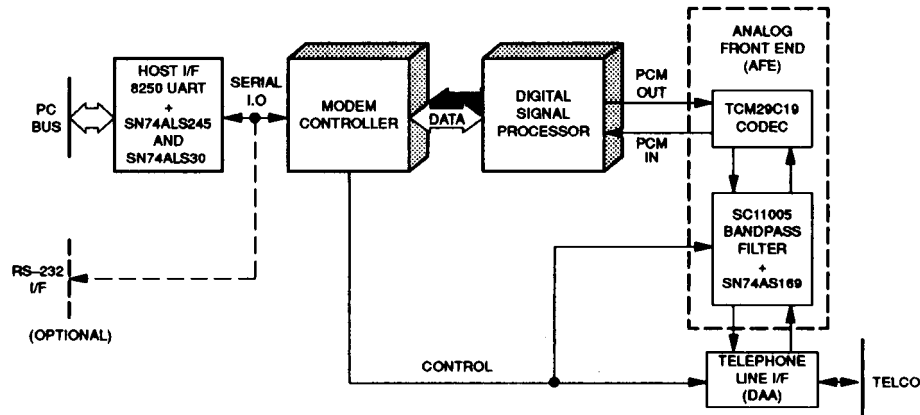


Figure 3. Block Diagram of Modem System Components

The designer must provide an interface between the host data terminal equipment and the modem controller. The DSP2400 uses an 8250 UART (plus a 74LS245 buffer and a 74ALS30 NAND Gate) to interface between a standard PC-AT and the modem controller. A standard RS-232C interface is used between the UART and the modem controller. The circuit diagram and additional information on the host interface used for the DSP2400 Modem can be found in [1].

The modem controller (80C51, TMS70C42, etc.) handles the overall modem control [3], directs the handshaking sequences, etc. It specifically performs the following functions:

1. AT command set interpretation
2. Scrambling/descrambling
3. Pulse dialing
4. Synchronous/asynchronous conversion
5. Modem configuration control
6. Protocol initialization

The modem controller sends a command to the DSP once per baud. Table 3 is a complete list of the commands, showing the structure and functions that are implemented.

Table 3. Modem Controller Commands for the DSP

Command	Code	Description
Protocol Select	Fxh	<p>Select protocol</p> <p>Bits 1, 0 – Speed select</p> <p>0 0 = 300 BPS</p> <p>0 1 = Reserved</p> <p>1 0 = Reserved</p> <p>1 1 = Reserved</p> <p>Bit 2 – CCITT/Bell</p> <p>0 = CCITT</p> <p>1 = Bell</p> <p>Bit 3 – Answer/originate</p> <p>0 = Answer</p> <p>1 = Originate</p>
Reserved	Exh	Reserved command
Operation Select	Dxh	<p>Select operating mode</p> <p>(bits 3, 2 reserved)</p> <p>0 0 = Line mode</p> <p>0 1 = Analog loopback</p> <p>1 0 = Reserved</p> <p>1 1 = Reserved</p>
Reserved	Cxh	Reserved command
Reserved	Bxh	Reserved command
Reserved	Axh	Reserved command
Transmit DTMF Tones	9xh	<p>Dial DTMF and return to configuration mode</p> <p>xxxx = D3-D0; numbers 0-9, A, B, C, D, *, and #</p>

Table 3. Modem Controller Commands for the DSP (Concluded)

Command	Code	Description
Transmit Mode Select	8xh	<p>Enable answer tone/data select</p> <p>Bits 1, 0 = Transmit select</p> <p>0 0 = Transmit idle</p> <p>0 1 = Transmit answer tone</p> <p>1 0 = Transmit data mode enable</p> <p>1 1 = Reserved</p> <p>Bits 3, 2 = Select answer tone frequency</p> <p>0 0 = 2100 Hz answer tone (V.21)</p> <p>0 1 = 2225 Hz answer mark (Bell 103)</p> <p>1 0 = 2025 Hz answer space (Bell 103)</p> <p>1 1 = Reserved</p>
Receive Mode Select	7xh	<p>Select receive configuration</p> <p>(bits 3,2 reserved)</p> <p>0 0 = Receive idle mode</p> <p>0 1 = Reserved</p> <p>1 0 = Receive data mode</p> <p>1 1 = Reserved</p>
Reserved	6xh	Reserved command
FSK Mode	5xh	<p>Select 300 BPS mode</p> <p>(bits 3,2,1 reserved)</p> <p>0 = 300 BPS mode deselect</p> <p>1 = 300 BPS mode select</p>
Reserved	4xh	Reserved command
Reserved	3xh	Reserved command
Reserved	2xh	Reserved command
Reserved	1xh	Reserved command
Reserved	0xh	Reserved command

As an example, the DSP2400 uses a masked ROM version of the TMS70C42 microcontroller (denoted as a TMS70C2400A) as the modem controller. The TMS70C2400A source code is available from Texas Instruments and includes provisions for the V.22bis and V.22 standard modems.

One noteworthy advantage of the TMS70C42/TMS320C17 interface is that it requires no external glue logic [7]. For complete information on the TMS70C2400 Modem Controller, including the call originate and answer sequences, refer to [2].

The TMS320 Digital Signal Processor performs the computationally intensive tasks such as modulation, demodulation, and tone generation and detection. It does not perform any control functions. Specifically, the TMS320 DSP performs the following functions:

1. Modulation/demodulation (V.21/Bell 103)
2. Data encoding/decoding
3. Filtering
4. Automatic gain control
5. Tone dialing
6. Call progress monitoring

The DSP is discussed in further detail in the next section of this application report. The DSP source code in Appendix B was originally part of the code developed for the TMS320A2400 Modem Digital Signal Processor (a ROM coded TMS320C17 DSP). The TMS320A2400 source code also includes V.22bis, V.22, and Bell 212A standard modems, with the software implementing the QAM and PSK modulation and demodulation schemes, carrier recovery, clock recovery, automatic gain control, and adaptive equalization functions. The TMS320A2400 and the source code is available from Texas Instruments.

Despite the differences between the code provided in Appendix B and the TMS320A2400 code, [1] and [3] are useful references, providing technical information about TMS320C17 modem applications.

The analog front end is composed of a TCM29C19 combo codec [9], a SC11005 bandpass filter [10] and a data access arrangement (DAA) telephone line interface composed of discrete components. The codec converts an 8-bit μ -law companded bit stream to an analog waveform and vice versa, at a 9.6-KHz sampling frequency. The SC11005 is a splitband filter that separates the transmit and receive carriers and performs the required signal shaping to the analog waveform. The DAA section is composed of a number of discrete components and is required to interface the modem to the public telephone network as dictated by FCC Rules Part 68. The analog front end circuit diagram is found in [1]. Further technical details are found in [2].

The DSP Software Implementation

The code provided in Appendix B is written specifically for a Texas Instruments TMS320C17 Digital Signal Processor. The key architectural features of the TMS320C17 are these:

1. 4 Kwords (8 Kbytes) of on-chip maskable ROM
2. 256 words of on-chip data RAM
3. Two full-duplex serial ports
4. On-chip companding hardware (μ - or A-Law)
5. On-chip sign magnitude/two's complement conversion hardware
6. A coprocessor port
7. 6.25-MIPS maximum execution speed

TMS320E17, with 4 Kwords of on-chip EPROM substituted for the 4 Kwords of maskable ROM, is also available for development and prototyping purposes. Refer to [8] and [11] for additional information on the TMS320C17 and TMS320E17.

The TMS320C17 source code listing file is found in Appendix B. The code requires approximately 50 words of data RAM and occupies 1100 words of program ROM. Of the 1100 words of program memory, 390 are coefficients, and the remaining 710 words are the program instructions. The software consists of a main program that references various subroutines. These are the main subroutines found in the program:

1. Command control interpreter (CCI)
2. FSK transmitter (FSKTX)
3. Dual-tone multifrequency transmitter (Part of FSKTX)
4. Automatic gain control (AGC)
5. FSK receiver (RSTSK)

The next section of text describes the main program. The subroutines are discussed in subsequent sections.

Figure 4 is a block diagram of the main program (code starting at beginning of main program label and ending at start of subroutines label) in Appendix B. Once the initialization of the data RAM and control registers (code beginning at start of additional tables label and ending at start of main program sequencer label) is complete, the main program loop is executed. The device remains in the WAIT loop (first four lines of code of main program sequencer routine) until the FR flag in the control register is raised. Control register bits 27-24 and 23-16 are set so the main program and data samples are transmitted/received to/from the TCM2919 codec at a rate of 9.6 KHz.

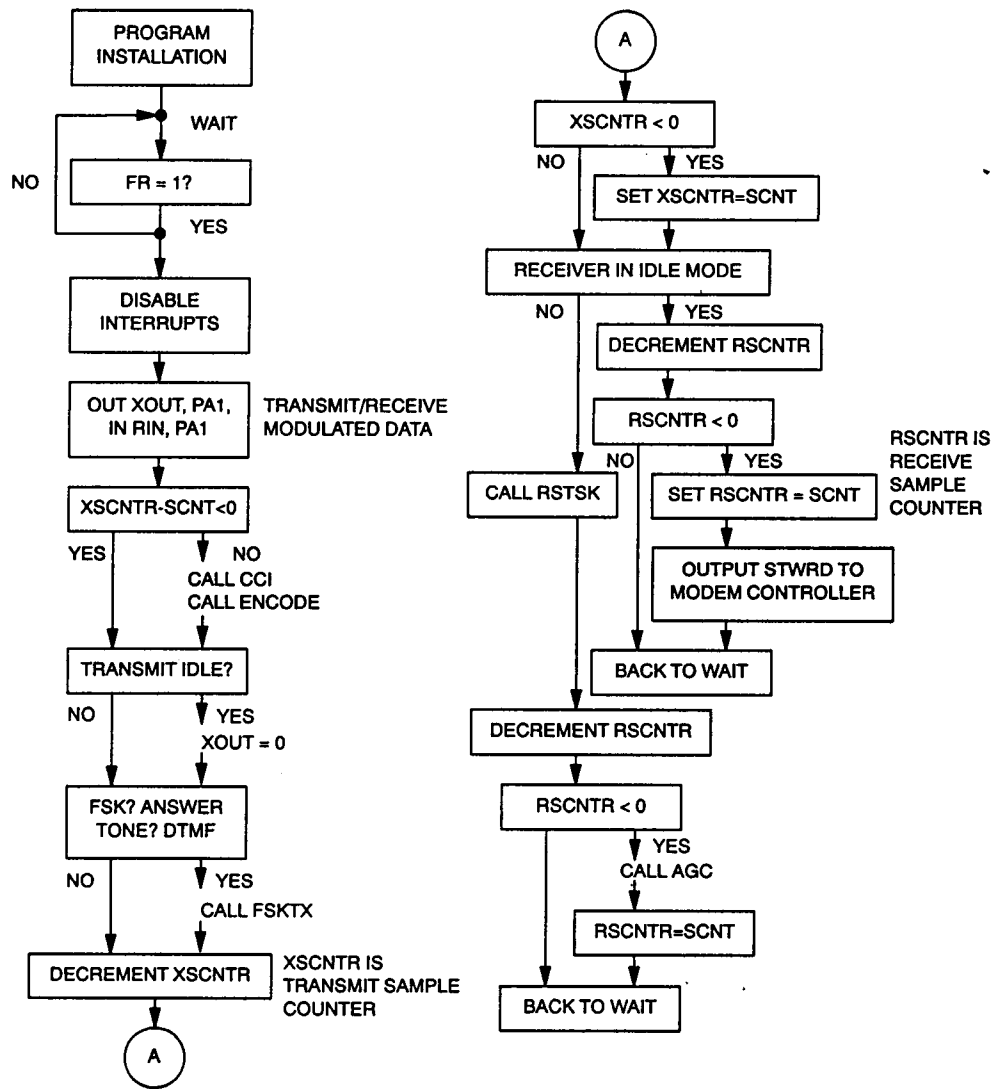


Figure 4. Flowchart of Main Program (Appendix B)

As the V.21/Bell 103 standard modems transmit data at 300 bps, a 9.6-KHz sampling rate results in 32 samples/ baud. The 9.6-KHz sampling rate is very practical for several reasons:

- It is higher than the Nyquist sampling frequency of approximately 8 KHz for a telephone channel, and
- It is a convenient multiple of the popular modem transmission frequencies (300, 1200, and 2400 bps).

The TMS320C17 is clocked by a 18.432-MHz oscillator. To satisfy the 9.6-KHz sampling frequency, the number of instructions executed per sample must be less than 480. To implement the various functions required by the FSK modulator/demodulator, it is necessary to distribute the tasks among the various samples within the baud. The command control interpreter (CCI) is executed during the first sample of the baud, and the AGC routine is implemented during the final sample baud.

When the raised FR flag is detected, the processor exits the WAIT loop and executes the main program. Refer to [8], Sections 3.8 and 3.9 for additional details on the FR flag, interrupts, and serial port. Table 4 describes the variables that are referenced in the main program.

Table 4. Variables Referenced in Main Program Variable

Variable Name	Description
XSCNTR	The transmit counter; equals the number of samples that have been transmitted in the current baud.
SCNT	Number of samples in a baud, i.e., 9.6 KHZ/300 HZ = 32 samples/baud.
XOUT	Output sample sent via the TX serial port to the combo codec.
RIN	Input sample sent via RX serial port from the combo codec.
STATUS	An 8-bit number used internally by the DSP. Indicates present operating mode of the modem.
STWRD	8-bit status word sent to the modem controller by the DSP. See Table 5.
OAFLAG	Indicates if modem is in originate or answer mode. OAFLAG = 0 → originate mode.
DTFLAG	Indicates if the modem is transmitting DTMF tones. DTFLAG = 1 → transmitting DTMF data.

Table 5 shows the organization of STWRD (the DSP status word that is written to the microcontroller).

Table 5 STWRD - DSP Status Word Written to the Modem Controller

Bit No.	Description
7	Enable/disable automatic gain control. 0 = Enable 1 = Disable
6	EDT (in band energy) 0 = Not detected 1 = Detected
5	Reserved
4	Reserved
3	Received data bit (0,1)
2	Reserved, set to 1
1	Reserved, set to 1
0	Reserved, set to 1

When the program exits the wait loop, it disables all interrupts and reads a data sample RIN from the receive buffer or writes a data sample XOUT to the transmit buffer of serial port #1.

At the first sample of a baud, when $XSCNTR = SCNT (=31)$, the program implements the command control interpreter (CCI) subroutine as shown in the following code. Note that $SCNT = 31$, and $XSCNTR$ is initially set at 31 and decremented by 1 every sample. When $XSCNTR$ equals 0, it is reset to 31, for a total of 32 samples.

```

        LAC    XSCNTR
        SUB    SCNT      ; ACCUM = XSCNTR-SCNT
        BLZ    SEQU      ; BRANCH TO SEQU IF ACCUM < 0
        CALL   CCI
SEQU:   LACK    030h

```

The CCI subroutine reads the next 8-bit command from the modem controller (TMS70C42400A or equivalent), performs the required program control functions, and returns to the main program.

If the DSP is in transmit idle mode, the data sample XOUT is set to 0 and sent to serial port #1 transmit buffer.

If the DSP is not in transmit idle, the FSK transmit subroutine FSKTX is called. Depending on the present value of STATUS as determined by the modem controller and the CCI subroutine, the FSKTX subroutine will transmit FSK encoded data, DTMF tones, or an answer tone. Upon completing the FSKTX subroutine, the program decrements the transmit sample counter XSCNTR by 1 and checks to see if it is less than 0. If so, XSCNTR is reset to 31. Otherwise, the program proceeds without any further modifications to XSCNTR.

At this point, the main program checks to see if the receiver is in idle mode. If the receiver is in idle mode, the receive sample counter RSCNTR is decremented. If RSCNTR is not less than 0, the program returns to the WAIT loop. If RSCNTR is now less than 0, it is reset to 31, and the program then returns to the WAIT loop.

If the receiver is not in idle mode, the receiver decode/demodulation subroutine RSTSK (receiver per sample task) is called. This subroutine demodulates the receiver signal and estimates the value of the received data. When the subroutine is complete, the main program decrements RSCNTR and resets it to 31, if required.

After the RSTSK subroutine is complete, the program decrements RSCNTR. If RSCNTR is greater or equal to 0, the program returns to the wait loop. For the sample, when RSCNTR is less than 0, the automatic gain control subroutine (AGC) is called once per baud. The AGC subroutine monitors and compensates for any significant variation of the received signal level caused by telephone line fluctuations and other dynamic effects. RSCNTR is then RESET to 31, and the program returns to the WAIT loop.

The main program calls the following subroutines:

- CCI—Command control interpreter
- DTMF—DTMF setup
- FSKSET—Set up FSK transmit frequency
- FSKTX—Transmitter mode select
- OPER—Set operating mode
- PROTO—Protocol select
- RESET—Reset and equalizer enable
- RMODE—Receiver mode select
- RSTSK—FSK demodulation
- XMODE—Transmitter mode select

Figure 5 shows a block diagram of the CCI subroutine. The CCI reads the setup command from the modem controller (through the co-processor port PA5) and stores it in data RAM location XDATA (The structure of XDATA is shown in Table 3). The CCI then calls the appropriate subroutine to modify the system control bits (OAFLAG and DTFLAG) and status register (STATUS). The CCI, depending whether the modem configuring the DSP is in answer, originate, or transmit DTMF, loads the required nominal frequency values into TXFRQ and RXFRQ. Table 6 shows the organization of the STATUS register.

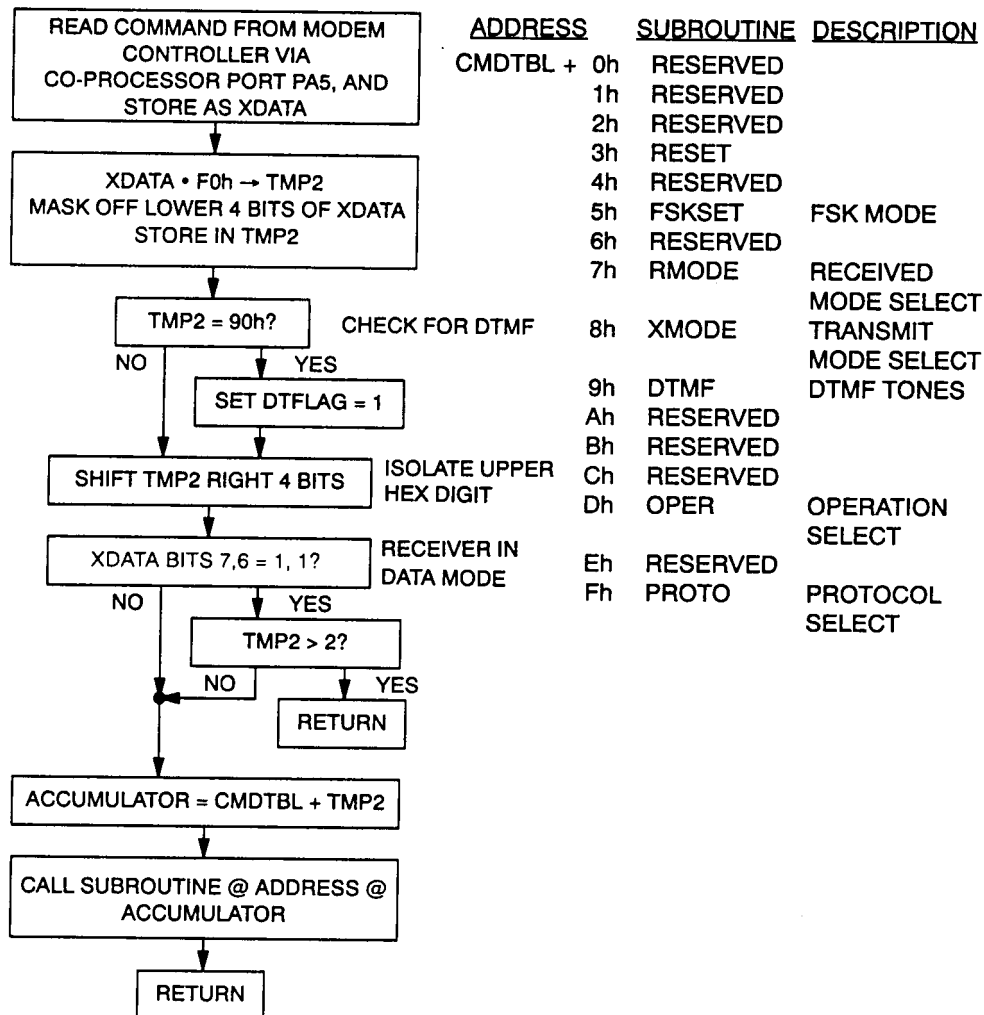


Figure 5. Flowchart of the CCI Subroutine

Table 6. The Status Register Organization

Bits	Description
7,6	Indicate Receiver Mode: 00 = Receiver in Idle Mode 01 = Call Progress Monitoring Mode 10 = Data Mode 11 = Reserved
5,4	Indicate Transmitter Mode: 00 = Transmitter in Idle Mode 01 = Transmit Answer Tone 10 = Data Mode 11 = Reserved
3	Answer/Originate Mode: 0 = Originate Mode 1 = Answer mode
2	CCITT/Bell Mode: 0 = CCITT (V.21) 1 = Bell (103)
1,0	Speed status: 00 = 300 BPS 01 = Reserved 10 = Reserved 11 = Reserved

The setup commands from the modem controller and subroutines called by the CCI subroutine are shown in Table 3.

The RESET subroutine loads 81h into the STWRD word that is sent to the modem controller via the co-processor port PA5. This advises the modem controller that the DSP has been reset. The DSP program then branches to START, and the DSP is reinitialized.

The FSKSET subroutine reads the XDATA word to determine if the next bit to be transmitted is 0 or 1 and then loads the appropriate 0 or 1 frequency F0ADD or F1ADD into the TXFRQ register.

When setup in answer mode, XDATA bits 3 and 2 are loaded into the STATUS register bits 7 and 6, respectively, by the RMODE subroutine. These bits determine what tasks the FSK receiver subroutine RSTSK will perform, as shown in Table 3 and Figure 5.

The XMODE subroutine reads XDATA bits 0 and 1. These bits determine what tasks the FSK transmitter subroutine FSKTX will perform as shown in Figure 4. If the transmit answer tone function is selected, bits 2 and 3 of XDATA indicate what the answer tone frequency will be:

XDATA Bits 3,2 =	0,0	2100 Hz
	0,1	2225 Hz
	1,0	Reserved
	1,1	Reserved

The program loads the appropriate answer tone value into register TXFRQ. XMODE then loads XDATA bits 1 and 0 into STATUS bits 5 and 4, respectively. STATUS bits 5 and 4 determine what tasks the transmitter subroutine FSKTX will perform.

The DTMF subroutine determines what number or symbol needs to be transmitted by reading XDATA bits 3 through 0. DTMF then loads the appropriate high-frequency phase step, low-frequency phase step, high-frequency gain, and low-frequency gain into the RXFRQ, TXFRQ, DTMFH, and DTMFL registers, respectively, from the Table TONTBL.

The OPER subroutine checks bits 1 and 0 of XDATA. If bits 1 and 0 equal 0 and 1 bit 3 of STATUS is set to 1, indicating that the modem is in analog loopback mode. If bits 1 and 0 are not equal to 0 and 1, OPER returns without performing any operations.

The PROTO subroutine selects the mode and protocol of the DSP based on XDATA bits 3 through 0. PROTO first sets bits 1 and 0 of STATUS (indicating the modem data rate), based on the value of bits 1 and 0 of XDATA (see Figure 7).

While the software provided in Appendix B supports only a 300-bps data rate, it does provide the necessary hooks so that different standard modems (ie V.22, V.22bis) can easily be incorporated into the code.

Next, PROTO checks XDATA bits 3 and 2 to determine if the modem should be in originate/answer mode and Bell/CCITT mode.

Bit 3:	0	= Originate
	1	= Answer
Bit 2:	0	= Bell
	1	= CCITT

As shown in Table 2, the transmission frequencies of the Bell 103 and V.21 originate and answer modes are unique. PROTO loads registers used by the FSK transmitter

subroutine (FSKTX) and the FSK receiver subroutine (RSTSK) with values stored in table TONTBL in data ROM and corresponding to transmit and receive frequencies.

PROTO then uses the XDATA bits 3 and 2 to determine which constants are transferred from table FSKTBL into addresses F1ADD (transmit 1 phase step), F0ADD (transmit 0 phase step), B1FSK (FSK delay filter coefficient), and GAIN (FSK mode gain). PROTO also loads addresses SCNT (baud counter=32), TRANS (FSK data transmission N=15), A1FSK (A1 demodulator filter coefficient), A2FSK (A2 demodulator filter coefficient), and DZONE (dead zone of window comparator) with the appropriate values.

If bit 3 of the STATUS word equals 1, the modem is set to analog loopback mode, and the modem should receive the information that it transmits. PROTO checks to see if bit 3 of STATUS equals 1; if so, the receiver parameters are modified to be the same band as the transmitter.

The FSK modulator is implemented in the FSKTX subroutine. Figure 6 is a block diagram of the FSKTX subroutine. The primary function of the FSK modulator is the following: Given a stream of binary data $a_0, a_1, a_2, \dots, a_{k-1}, a_k$ for each data element $a_k = \{0,1\}$, generate a corresponding signal of frequency f_0 or f_1 for the duration of a_k 's baud period.

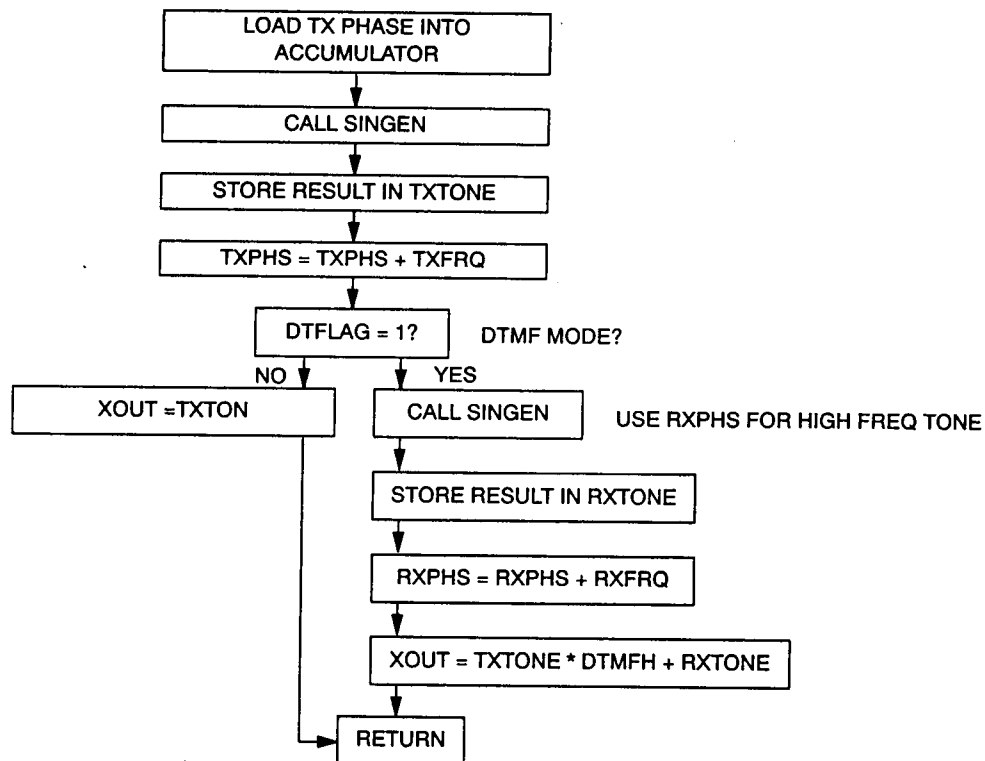


Figure 6. Flowchart of Subroutine FSKTX

Figure 7 shows a functional model of the FSK modulator. The TMS320 software implementation of the FSK modulator generates tones by stepping through a cosine table. The size of the phase step determines the output signal frequency. You should pay particular attention how phase angles, phase steps, cosines, and sines are represented as 16- and 32-bit integer numbers.

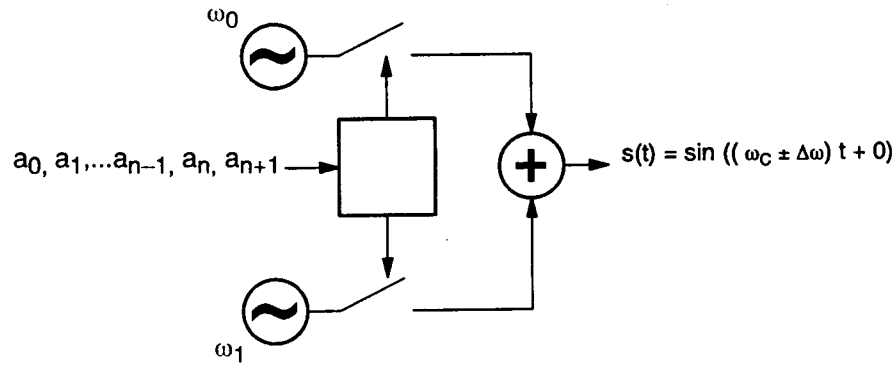


Figure 7. Functional Model of an FSK Modulator

Table 7 describes the significant variables used in the FSKTX subroutine.

Table 7. Variables Referenced in the FSK Transmitter Subroutine FSKTX

Variable Name	Description
TXPHS	Present value of the transmit signal phase. Also used as present phase of the low frequency DTMF tones.
TXFRQ	Phase step between consecutive TXPHS samples.
RXPHS	Normally used in the FSK demodulator subroutine RXTSK. Used as present phase for the high frequency DTMF tone.
RXFRQ	Normally used in RXTSK subroutine. Also used as phase step for high-frequency tone when transmitting DTMF tones.
DTMFL	Scaling factor for low-frequency DTMF tones.
DTMFH	Scaling factor for high-frequency DTMF tones.
SINGEN	A subroutine called by FSKTX. Given a 16-bit number representing an angle from 0 to Pi, the SINGEN routine determines the sine of the angle and stores the result at address TMP3.

The software FSK Modulation routine receives data at a rate of 300 bps and generates 12-bit, two's complement data samples at a rate of 9.6 KHz. The TMS320C17's on-chip hardware compander reduces the sample to 8 bits before it is sent to the Codec via the serial port.

The most recent phase of the output signal is stored in data memory location TXPHS, and the amplitude is read from the COSOFF table by the SINGEN subroutine. The frequency of the transmitted signal is determined by the size of the phase step TXFRQ between successive output samples:

$$\text{TXPHS}[(N+1)T] = \text{TXPHS}[NT] + \text{TXFRQ}[NT]$$

The value of TXFRQ is determined by the FSKSET subroutine referenced by the CCI subroutine. Recall that, depending on the instruction received from the modem controller at the beginning of the baud, the CCI subroutine loaded data memory location TXPHS with either F0ADD or F1ADD. Table 8 shows the FSK frequencies and phase steps (TXFRQ) for the V.21 and Bell 103 modem standards.

Table 8. Frequencies and Phase steps for V.21 and Bell 103 Modems

Modem Standard		Frequency (Hz)	Phase Step @9.6 KHz	Phase Step TXFRQ, Q15 hex
V.21	Originate 1	980	$0.2042 \cdot \pi$	1A22h
	Originate 0	1180	$0.2458 \cdot \pi$	1F77h
	Answer 1	1650	$0.3428 \cdot \pi$	2C00h
	Answer 0	1850	$0.3854 \cdot \pi$	3155h
Bell 103	Originate 1	1270	$0.2646 \cdot \pi$	21DDh
	Originate 0	1070	$0.2229 \cdot \pi$	1C89h
	Answer 1	2225	$0.4635 \cdot \pi$	3B55h
	Answer 0	2025	$0.4219 \cdot \pi$	3600h

The magnitude of the phase step is determined by

$$[(\text{Desired Frequency})/(\text{Sampling Frequency})] * 2\pi$$

In the case of the originate 1 of the V.21 modem, the phase step equals

$$(1270/9600) * 2\pi = .2646 \pi \text{ Radians}$$

Both TXPHS and TXFRQ data memory locations are 16-bit binary numbers in Q15 two's complement notation equal to

$$(\text{Output Signal Phase})/\pi.$$

Thus TXPHS hex values

$$\begin{aligned} 2000h &= \pi/4 \\ 4000h &= \pi/2 \\ 6000h &= 3\pi/4 \\ 8000h &= -\pi \\ A000h &= -3\pi/4 \end{aligned}$$

An advantage of this approach is that the phase of the output signal is continuous. This provides a higher spectral efficiency than that of a discontinuous phase FSK implementation.

The sine generation subroutine SINGEN subtracts $\pi/2$ (4000h) from TXPHS and uses this phase to read the amplitude from the COSOFF table. The symmetry of the cosine function has been used to reduce the table size from 513 to 257 elements, with data memory addresses COSOFF, COSOFF + 128, and COSOFF + 256 corresponding to 0, $\pi/2$, and π radians, respectively. To determine the cosine of an angle outside the 0-to- π range, the program utilizes the two's complement format of the data and the absolute value function ABS. As an example, assume that the present phase TXPHS is

$$\text{TXPHS}(N) = (-170/256) * \pi = -.6640625 * \pi = A600h$$

If we are transmitting a 1 in V.21 Originate mode, the phase step is

$$\text{TXFRQ} = .26448 * \pi = 21DDh$$

The next value of:

$$\begin{aligned} \text{TXPHS}(N+1) &= \text{TXPHS}(N) + \text{TXFRQ} \\ &= -.6640625 \pi + .26448 \pi \\ &= -.3995825 \pi \\ &= A600h + 21DDh = C7DDh \end{aligned}$$

The subroutine then subtracts $\pi/2$ (4000h) from TXPHS, so the sine of angle TXPHS can be determined from the Cosine table:

$$\begin{aligned} \text{ANGLE} &= \text{TXPHS}(N+1) - \pi/2 \\ &= -.3995825 \pi - .5 \pi \\ &= -.8995825 \pi \\ &= C7DDh - 4000h = 87DDh \end{aligned}$$

Note that TXRFQ is added to TXPHS(N), and $\pi/2$ is subtracted from TXPHS(N+1) with the sign extension suppressed, so TXPHS(N+1) = 87DDh. This represents 1.06143 π as an unsigned number or $-.93857 \pi$ as a signed number. If we now consider TXPHS(N+1) a signed and take the absolute value:

$$\text{ABS}[\text{TXPHS}] = \text{ABS}[87DDh] = 7823h \text{ representing } .93857 \pi$$

Note that:

$$\cos(1.06143\pi) = \cos(.93857\pi) = -.98144$$

The cosine table address is generated:

$$\text{COSOFF} + (7823\text{h}/80\text{h}) = \text{COSOFF} + \text{F0h}$$

The value at Data Memory address COSOFF + F0h is

$$\cos((240/256)\pi) = -.980786 = 8276\text{h}, \text{Q15 2's complement notation}$$

Within the limits of the cosine table precision, the calculated output value equals the value read from the table.

The structure of the FSK Demodulator is shown in Figure 8.

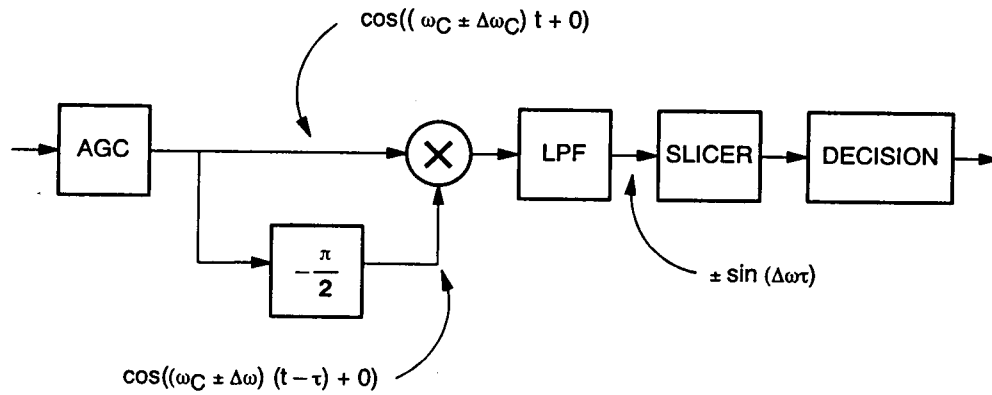


Figure 8. FSK Demodulator

The received FSK signal is sent to the DSP from the Codec via the serial port. The on-chip companding hardware expands the signal from an 8- to 13-bit value. The automatic gain control routine compensates for transient signal level variations and sends the amplitude adjusted received signal $R(t)$ to the software demodulator.

$$R(t) = \cos[(\omega_c \pm \delta\omega) * t + \phi] \quad (4)$$

As this is a binary FSK system, the frequency of this signal is either $\omega_c - \delta\omega$ or $\omega_c + \delta\omega$, depending on whether a 0 or 1 was sent. (Recall from the V.21 signal that $\delta\omega$ is less than 0.)

The received signal $R(t)$ is multiplied by a delayed version of itself:

$$R(t - \tau) = \cos[(\omega_c \pm \delta\omega) * (t - \tau) + \phi] \quad (5)$$

Where τ is the signal delay.

The product of the received signal (4) and delayed received signal (5) is

$$2 * \cos[(\omega_c \pm \delta\omega) * t + \phi] * \cos[(\omega_c \pm \delta\omega) * (t - \tau)] \quad (6)$$

$$= \cos[2(\omega_c \pm \delta\omega) * t - (\omega_c \pm \delta\omega) * \tau + 2 * \phi] + \cos[(\omega_c \pm \delta\omega) * \tau] \quad (7)$$

If $\omega_c \tau$ is set to equal $\pi/2$, and (7) is lowpass filtered to remove the double frequency component, the resulting signal is

$$\cos(\pi/2 \pm \delta\omega * \tau) = \sin(\pm \delta\omega \tau) = \pm \sin(\delta\omega) \quad (8)$$

If $\delta\omega$ is greater than 0, then the sign of the lowpass filter output will be positive or negative, depending on whether $\omega_c + \delta\omega$ or $\omega_c - \delta\omega$ is originally transmitted. If $\delta\omega$ is less than 0, obviously the opposite relationship is true. The sign of the lowpass filter output indicates the value of the received data.

The TMS320 software implementation of the 300-bps FSK Demodulator is found in Subroutine RSTSK, Subroutine CCITT, and Subroutine FDEM20 in Appendix B.

The AGC subroutine provides the RSTSK subroutine with a Q11 two's complement format received signal sample at a rate of 9.6 K samples per second.

As previously discussed, the data is extracted from the received signal by multiplying the received signal by a $\pi/2$ delayed version of itself, $\cos[(\omega_c \pm \delta\omega) * t + \phi - \pi/2 \pm \delta\omega * \tau]$. The product is then passed through a lowpass filter to remove the high frequency components.

If the desired phase delay is

$$\omega_c * \tau = \pi/2, \quad (9)$$

then

$$\tau = 1/(4 * f_c) \quad (10)$$

The sample rate is 9.6 KHz, or a period $T = 104.167 \mu s$. Table 9 shows the carrier frequencies, for both the V.21 and Bell 103 standards, the time delays corresponding to a $\pi/2$ phase delay and the equivalent number of 9.6-Khz samples. Note that none of the delays are exact multiples of the 9.6-KHz sampling period; each delay has an integer and fractional part.

Table 9. Carrier Frequency and Time Delays

Modem Standard		Frequency (Hz)	$\tau(\mu s)$	# of 9.6-KHz Samples
V.21	Originate	1080	231.481	2.2222
	Answer	1750	142.857	1.3714
Bell 103	Originate	1170	213.675	2.0513
	Answer	2125	117.647	1.1294

To minimize the probability of error, it is necessary that the phase delay be as close to $\pi/2$ as possible. An accurate estimate of the fractional part of the delay must be total phase delay. This is achieved by using a single zero FIR filter.

$$R((n - \alpha)T) = \text{GAIN} * [R(nT) + \text{B1FSK} * R((n-1)T)] \quad (11)$$

where $R(nT)$ is the n th sample of the received signal $R(t)$

$R((n - \alpha)T)$ is the estimate of the fractionally delayed signal

n is an integer

α is the desired fractional delay, $0 < \alpha < 1$

The filter coefficient B1FSK and GAIN for the fractional delay filter of each V.21 and Bell 103 carrier are shown in Table 10. The derivation of the gain and filter coefficients are shown in Appendix A.

Table 10. Time Delay and FIR Filter Coefficients

Modem Standard		Frequency	Fractional Delay 9.6-KHz Sample(____)	Gain	B1FSK
V.21	Originate	1080	.2222	.69753	.32796
	Answer	1750	.3714	1.00000	.68889
Bell 103	Originate	1170	.0518	.57731	.07175
	Answer	2125	.1294	1.00000	.31678

B1 and GAIN are stored in data memory locations B1FSK and GAIN, respectively. The actual implementation is

$$\text{PDEL1} = \text{AGCOUT} + \text{B1FSK} * \text{PDEL0}$$

where AGCOUT is the received signal after the signal level has been compensated by the automatic gain control routine.

$$\begin{aligned}
\text{AGCOUT} &= \cos[(\omega_c \pm \delta\omega) * nT + \phi] \\
\text{PDEL0} &= \cos[(\omega_c \pm \delta\omega) * (n-1)T + \phi] \\
\text{PDEL1} &= \cos[(\omega_c \pm \delta\omega) * (n-1-\alpha)T + \phi], \quad 0 < \alpha < 1 \\
\text{PDEL2} &= \cos[(\omega_c \pm \delta\omega) * (n-2-\alpha)T + \phi]
\end{aligned}$$

Since AGCOUT, PDEL0, PDEL1, and PDEL2 are consecutive data memory locations, the integer multiples of the 9.6-KHz sample delays are easily achieved by using the data move (DMOV) instruction. PDEL1 is calculated after the demodulator product operation and is not used until the next sample period, a delay of one sample period.

For the low-frequency carriers of the V.21 and Bell 103 standards, a second delay is required and is implemented as DMOV PDEL1, moving the contents of PDEL1 into data memory PDEL2.

When the sample delayed signal (PDEL1 or PDEL2 for the high- or low-frequency carriers, respectively) is generated, it is multiplied by the most recent sample AGCOUT. The product of the multiply is stored in data memory location PROD. PROD is multiplied by GAIN and then filtered by a second-order direct-form, lowpass IIR filter, and the result is stored in location LPFOUT. Further information on digital filters can be found in [12], [13].

Given the lowpass filter output LPFOUT, the FSK demodulator must now estimate the value of the received signal.

In the Data Estimation routine, the following memory location addresses are called:

- BDATA** — The data estimation for the previous baud.
- FSKDAT** — Data estimation of the current sample.
- BAUDCK** — A record of the number of samples presently taken in the current baud. Recall that the sample rate is 9.6 KHz and the baud rate is 300 Hz; so there are 32 samples/baud.
- COUNTR** — The data estimations of each sample in the current baud are compared to the decision of the previous baud. If these are different, then COUNTR is incremented. If COUNTR reaches 32 before BAUDCK reaches 32, it is assumed that a data transition has occurred, and BDATA is set to the opposite value:

$$\text{BDATA}(N+1) = \text{ABS}[\text{BDATA}(N) - 1]$$

Figure 9 is a flowchart of the data decision source code implementation.

AGCOUT - RECEIVED SAMPLE $t = nT$
 PDEL0 - AGCOUT @ $t = (n-1)T$
 PDEL1 - AGCOUT + BI + PDEL0
 PDEL2 - PDEL1 @ $t = (n-1)T$

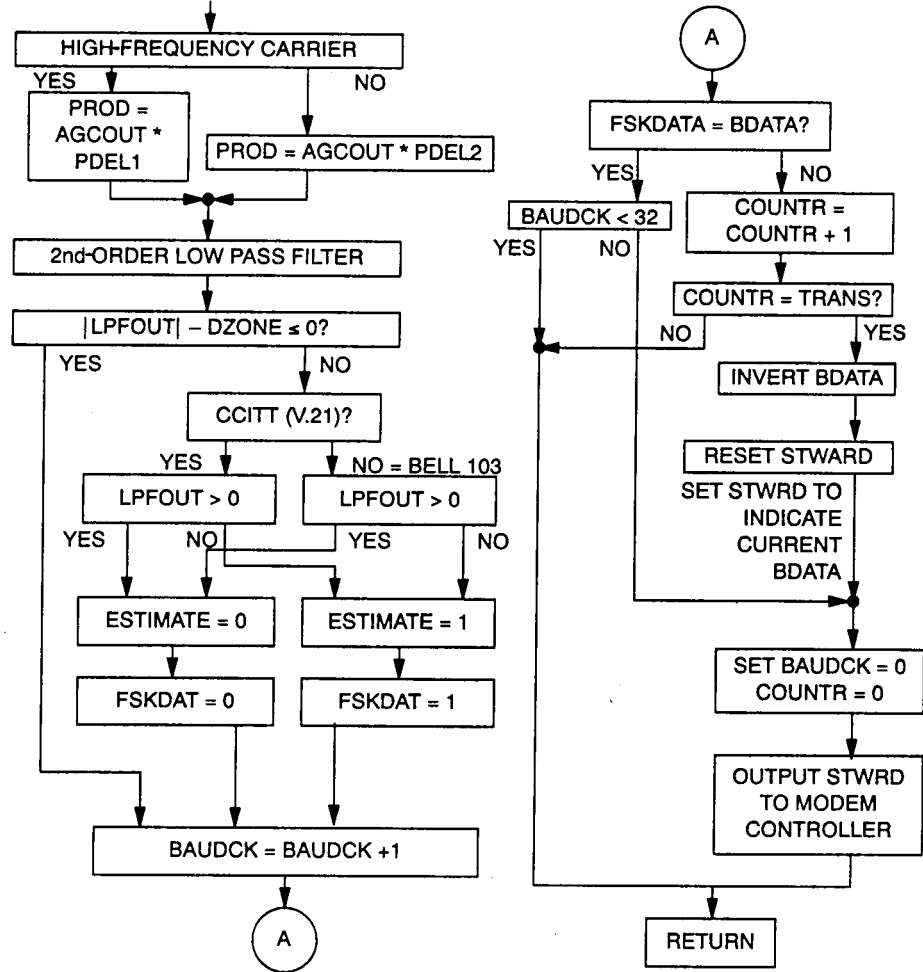


Figure 9. Data Decision Algorithm Flowchart

The function of the automatic gain control subroutine AGC is to compensate for amplitude distortions introduced by the telephone system, etc. References [5], [14] provide additional information on AGC.

Incorporating Additional Functions into the DSP

One of the important tasks the designer faces is incorporating value-added software functions into the DSP source code found in Appendix B.

The software presented here uses only 1.1 Kwords of the 4 Kwords of maskable ROM available on the TMS320C17. This provides you with a significant amount of code space to implement value-added functions.

This software offers a number of hooks that facilitate the easy inclusion of additional software. Note in Table 3 (Modem Controller Commands for the DSP), that the following commands are presently reserved : E, C, B, A, 6, 4, 2, 1, and 0. Each of these commands have bits 0 through 3 undefined. All of these commands can be used by the designer to call additional functions.

You must ensure that the correct modifications are made to the modem controller and modem DSP software. The DSP control command interpreter (CCI) must be modified to recognize and respond to the new commands. The additional functions should be implemented in either a new or the appropriate existing subroutine. The option indicating to the main program that the new subroutine should be called, needs to be provided. This can be done using the STATUS register, or you can define a new register.

You must also ensure that the XDATA word will indicate the present status of the DSP to the modem controller. There are presently a number of unused bits in the XDATA word, so incorporating the modifications in the DSP is straightforward.

Finally, you must ensure that the additional software functions do not exceed the timing requirements imposed by the 9600-KHz sampling frequency.

Conclusions

This application report presented you with the information required to implement a 300-bps V.21/Bell 103 FSK modem based on a TMS320C17 Digital Signal Processor. Both hardware and software issues were discussed. A summary of the FSK modulation and demodulation algorithms and a basic review of modems were also provided. A discussion about incorporation of additional functions and software into the code provided concluded this report.

Appendix A is a derivation of the FSK demodulator fractional delay filter coefficients. Appendix B is the TMS320C17 source code listing.

Acknowledgements

The author wishes to acknowledge the contribution of Dr. Amin Haoni of Technekron, Inc., and George Troullinos, and Raj Chirayil of Texas Instruments. This report is based on their work.

Glossary of Symbols and Abbreviations

- bps — Bits per second
- FSK — Frequency shift keying
- ω_c — Carrier signal angular velocity
- $\delta\omega$ — Modulation shift of angular velocity
- t — Time
- ϕ — Phase shift
- ω_0 — Angular velocity transmitted to indicate a 0
- ω_1 — Angular velocity transmitted to indicate a 1
- τ — The amount of time the received signal is delayed in the FSK demodulator
- f_0 — Frequency transmitted to indicate a 0
- f_1 — Frequency transmitted to indicate a 1
- f_c — Carrier frequency
- α — Sample fractional delay created by the single FIR filter

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Appendix A

Calculation of Phase Delay Filter Coefficients

A key element of the FSK demodulator implementation is the $\pi/2$ phase delay of the carrier signal. The effectiveness of the demodulator is highly dependent on the accuracy of the $\pi/2$ phase delay.

In a digital system, it is highly unlikely that the time delay required for the $\pi/2$ phase delay is an exact multiple of the signal sampling period. It will be necessary to introduce phase delays that are a fraction of the sampling period.

To accurately generate the fractional delay, the digital signal processor uses a single zero FIR filter. This appendix derives the coefficients for the single zero FIR.

Given the one zero FIR filter shown in Figure A-1:

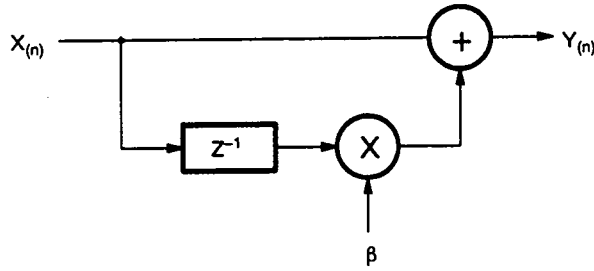


Figure A-1. One Zero FIR Filter.

$$Y(n) = X(n) + \beta X(n-1) \quad (A1)$$

therefore

$$\begin{aligned} Y(z) &= X(z) + \beta * z^{-1} X(z) \\ &= X(z) * (1 + \beta z^{-1}) \end{aligned} \quad (A2)$$

The transform of the filter is $F(z)$

$$F(z) = Y(z)/X(z) = (1 + \beta z^{-1}) \quad (A3)$$

The purpose of this filter is to introduce a precise group delay τ (delay of the signal envelope) to the received signal τ . is defined as

$$\tau = \frac{-d\theta(\omega)}{d\omega} \Delta = \text{group delay} \quad (A4)$$

Evaluate $F(z)$ at $z = e^{j\omega}$ to obtain the frequency response.

$$F'(\omega) = F(e^{j\omega}) = 1 + \beta e^{-j\omega} \quad (A5)$$

$$F'(\omega) = R(\omega) + jI(\omega) = A(\omega)e^{j\phi(\omega)} \quad (A6)$$

Where $R(\omega)$, $I(\omega)$, $A(\omega)$, and $\phi(\omega)$ are real functions of ω .

$$A(\omega) = |F'(\omega)| = [R(\omega)^2 + I(\omega)^2]^{1/2} \quad (A7)$$

and

$$\phi(\omega) = \arctan(I(\omega)/R(\omega)) \quad (A8)$$

Given

$$e^{-j\omega} = \cos\omega - j\sin\omega \quad (A9)$$

Substituting (A9) into (A5)

$$F'(\omega) = 1 + \beta \cos\omega - j\beta \sin\omega \quad (A10)$$

From (A6), (A8), and (A10)

$$\phi(\omega) = \left(\arctan \frac{-\beta \sin\omega}{1 + \beta \cos\omega} \right) \quad (A11)$$

Substituting (A11) into (A5)

$$\frac{d\phi(\omega)}{d\omega} = \frac{d}{d\omega} \left(\arctan \left(\frac{-\beta \sin\omega}{1 + \beta \cos\omega} \right) \right) \quad (A12)$$

now

$$\frac{d}{dx} (\arctan(u)) = \frac{1}{1 + u^2} * \frac{du}{dx} \quad (A13)$$

therefore

$$\tau = \frac{-1}{1 + \left(\frac{-\beta \sin\omega}{1 + \beta \cos\omega} \right)^2} * \frac{d}{d\omega} \left(\frac{-\beta \sin\omega}{1 + \beta \cos\omega} \right) \quad (A14)$$

$$\begin{aligned} &= \left(\frac{-(1 + \beta \cos\omega)^2}{1 + \beta^2 + 2\beta \cos\omega} \right) * \left(\frac{-\beta \cos\omega - \beta}{(1 + \beta \cos\omega)^2} \right) \\ &= \frac{+\beta(\beta + \cos\omega)}{1 + \beta^2 + 2\beta \cos\omega} \end{aligned} \quad (A15)$$

Assuming τ is expressed in terms of sample delays D

$$D = \frac{+ \beta (\beta + \cos\omega)}{(1 + \beta^2 + 2 \beta \cos\omega)} \quad (\text{A17})$$

Rearranging (A17) and using the quadratic equation to solve for

$$\beta = - \frac{(1-2D)\cos\omega \pm ((1-2D)^2\cos^2\omega + 4D(1-D))^{1/2}}{2(1-D)} \quad (\text{A18})$$

Given the desired group delay D, and the frequency $f = \omega/2\pi$ (where $\omega = (1080/9600)2\pi$), the filter coefficient β can be determined using equation (A18).

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*****
*
* APPENDIX B
*
*****
*
* V21/BELL 103 MODEN
*
*****
*
* ASSEMBLY LANGUAGE SOFTWARE FOR TMS320C17 IMPLEMENTATION.
*
* ALL RIGHTS RESERVED BY TEXAS INSTRUMENTS (C)
*
*****
*
* VERSION 1.0 01/SEP/89
*
*****
*
* .option x
*
*****
*
* CONSTANT DEFINITIONS
*
*****
*
* RNSK1: .set 000h
* RNSK2: .set 00h
* LNSK1: .set 0F3h
* LNSK2: .set 0Ah
* INTRSK1: .set 00Fh
* INTRSK2: .set 00Fh
* ASPRSK1: .set 0FCh
* TSKSH: .set 0Ah
* RCSH: .set 0Ah
*
*****
*
* ACC EQUATES
*
*****
*
* ACCDEF: .set 0586h
*
*****
*
* HIGH PASS FILTER CONSTANT
*
*****
*
* HPAU: .set 14 ; #16 - 14
*
*****
*****
*
* DATA MEMORY (RAM) ASSIGNMENTS
*
*****
*
* STATUS: .set 0
* ADDR4: .set 1
* DFLAG: .set 2
* ONE: .set 3
* SL010: .set 16
* STMR0: .set ONE+1
* XSNTR: .set STMR0+1
* RSNTR: .set XSNTR+1
* SNT: .set RSNTR+1
* XRTS: .set SNT+1
* XRT: .set XRTS+1
* RIN: .set XRT+1
* TMRQ: .set RIN+1
* TPRS: .set TMRQ+1
* RFRQ: .set TPRS+1
* RFRSH: .set RFRQ+1
* TR4: .set RFRSH+1
* TR5: .set TR4+1
*
*****
*
* REC. BAUD COUNTER
*
*****
*
* NORMAL BAUD COUNTER
*
*****
*
* OUTPUT SAMPLE
*
*****
*
* INPUT SAMPLE
*
*****
*
* RECEIVE DEMODULATION ANGLE (KH)
*
*****
*
* PARTIAL FILTERED SIGNAL w(n)
*
*****
*
* TMR5+1
* TMR2: .set TMR5+1
* TMR0: .set TMR2+1
*
*****
*
* TMR3: .set TMR0+1
* TMR3+1: .set TMR3+1
*
*****
*
* TMR1+1: .set TMR3+1
* TMR1: .set TMR1+1
* TMR0: .set TMR1+1
*
*****
*
* TMR0: .set TMR0+1
*
*****
*
* IN THE FSK MODE, THE T1/R1 PARAMETERS SHARE THE SAME MEMORY AS THE
*
*****
*
* EQUALIZER DELAY LINE
*
*****
*
* Q0: .set IR0
* ACOUT: .set PROD+1
* PRL0: .set ACOUT+1
* PRL1: .set PRL0+1
* PRL2: .set PRL1+1
* PRL3: .set PRL2+1
* PRL4: .set PRL3+1
* PRL5: .set PRL4+1
* PRL6: .set PRL5+1
* PRL7: .set PRL6+1
* PRL8: .set PRL7+1
* PRL9: .set PRL8+1
* PRL10: .set PRL9+1
* PRL11: .set PRL10+1
* PRL12: .set PRL11+1
* PRL13: .set PRL12+1
* PRL14: .set PRL13+1
* PRL15: .set PRL14+1
* PRL16: .set PRL15+1
* PRL17: .set PRL16+1
* PRL18: .set PRL17+1
* PRL19: .set PRL18+1
* PRL20: .set PRL19+1
* PRL21: .set PRL20+1
* PRL22: .set PRL21+1
* PRL23: .set PRL22+1
* PRL24: .set PRL23+1
* PRL25: .set PRL24+1
* PRL26: .set PRL25+1
* PRL27: .set PRL26+1
* PRL28: .set PRL27+1
* PRL29: .set PRL28+1
* PRL30: .set PRL29+1
* PRL31: .set PRL30+1
* PRL32: .set PRL31+1
* PRL33: .set PRL32+1
* PRL34: .set PRL33+1
* PRL35: .set PRL34+1
* PRL36: .set PRL35+1
* PRL37: .set PRL36+1
* PRL38: .set PRL37+1
* PRL39: .set PRL38+1
* PRL40: .set PRL39+1
* PRL41: .set PRL40+1
* PRL42: .set PRL41+1
* PRL43: .set PRL42+1
* PRL44: .set PRL43+1
* PRL45: .set PRL44+1
* PRL46: .set PRL45+1
* PRL47: .set PRL46+1
* PRL48: .set PRL47+1
* PRL49: .set PRL48+1
* PRL50: .set PRL49+1
* PRL51: .set PRL50+1
* PRL52: .set PRL51+1
* PRL53: .set PRL52+1
* PRL54: .set PRL53+1
* PRL55: .set PRL54+1
* PRL56: .set PRL55+1
* PRL57: .set PRL56+1
* PRL58: .set PRL57+1
* PRL59: .set PRL58+1
* PRL60: .set PRL59+1
* PRL61: .set PRL60+1
* PRL62: .set PRL61+1
* PRL63: .set PRL62+1
* PRL64: .set PRL63+1
* PRL65: .set PRL64+1
* PRL66: .set PRL65+1
* PRL67: .set PRL66+1
* PRL68: .set PRL67+1
* PRL69: .set PRL68+1
* PRL70: .set PRL69+1
* PRL71: .set PRL70+1
* PRL72: .set PRL71+1
* PRL73: .set PRL72+1
* PRL74: .set PRL73+1
* PRL75: .set PRL74+1
* PRL76: .set PRL75+1
* PRL77: .set PRL76+1
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* PRL82: .set PRL81+1
* PRL83: .set PRL82+1
* PRL84: .set PRL83+1
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* PRL86: .set PRL85+1
* PRL87: .set PRL86+1
* PRL88: .set PRL87+1
* PRL89: .set PRL88+1
* PRL90: .set PRL89+1
* PRL91: .set PRL90+1
* PRL92: .set PRL91+1
* PRL93: .set PRL92+1
* PRL94: .set PRL93+1
* PRL95: .set PRL94+1
* PRL96: .set PRL95+1
* PRL97: .set PRL96+1
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* PRL100: .set PRL99+1
* PRL101: .set PRL100+1
* PRL102: .set PRL101+1
* PRL103: .set PRL102+1
* PRL104: .set PRL103+1
* PRL105: .set PRL104+1
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* PRL107: .set PRL106+1
* PRL108: .set PRL107+1
* PRL109: .set PRL108+1
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* PRL111: .set PRL110+1
* PRL112: .set PRL111+1
* PRL113: .set PRL112+1
* PRL114: .set PRL113+1
* PRL115: .set PRL114+1
* PRL116: .set PRL115+1
* PRL117: .set PRL116+1
* PRL118: .set PRL117+1
* PRL119: .set PRL118+1
* PRL120: .set PRL119+1
* PRL121: .set PRL120+1
* PRL122: .set PRL121+1
* PRL123: .set PRL122+1
* PRL124: .set PRL123+1
* PRL125: .set PRL124+1
* PRL126: .set PRL125+1
* PRL127: .set PRL126+1
* PRL128: .set PRL127+1
* PRL129: .set PRL128+1
* PRL130: .set PRL129+1
* PRL131: .set PRL130+1
* PRL132: .set PRL131+1
* PRL133: .set PRL132+1
* PRL134: .set PRL133+1
* PRL135: .set PRL134+1
* PRL136: .set PRL135+1
* PRL137: .set PRL136+1
* PRL138: .set PRL137+1
* PRL139: .set PRL138+1
* PRL140: .set PRL139+1
* PRL141: .set PRL140+1
* PRL142: .set PRL141+1
* PRL143: .set PRL142+1
* PRL144: .set PRL143+1
* PRL145: .set PRL144+1
* PRL146: .set PRL145+1
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* PRL149: .set PRL148+1
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* PRL151: .set PRL150+1
* PRL152: .set PRL151+1
* PRL153: .set PRL152+1
* PRL154: .set PRL153+1
* PRL155: .set PRL154+1
* PRL156: .set PRL155+1
* PRL157: .set PRL156+1
* PRL158: .set PRL157+1
* PRL159: .set PRL158+1
* PRL160: .set PRL159+1
* PRL161: .set PRL160+1
* PRL162: .set PRL161+1
* PRL163: .set PRL162+1
* PRL164: .set PRL163+1
* PRL165: .set PRL164+1
* PRL166: .set PRL165+1
* PRL167: .set PRL166+1
* PRL168: .set PRL167+1
* PRL169: .set PRL168+1
* PRL170: .set PRL169+1
* PRL171: .set PRL170+1
* PRL172: .set PRL171+1
* PRL173: .set PRL172+1
* PRL174: .set PRL173+1
* PRL175: .set PRL174+1
* PRL176: .set PRL175+1
* PRL177: .set PRL176+1
* PRL178: .set PRL177+1
* PRL179: .set PRL178+1
* PRL180: .set PRL179+1
* PRL181: .set PRL180+1
* PRL182: .set PRL181+1
* PRL183: .set PRL182+1
* PRL184: .set PRL183+1
* PRL185: .set PRL184+1
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* PRL189: .set PRL188+1
* PRL190: .set PRL189+1
* PRL191: .set PRL190+1
* PRL192: .set PRL191+1
* PRL193: .set PRL192+1
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* PRL195: .set PRL194+1
* PRL196: .set PRL195+1
* PRL197: .set PRL196+1
* PRL198: .set PRL197+1
* PRL199: .set PRL198+1
* PRL200: .set PRL199+1
* PRL201: .set PRL200+1
* PRL202: .set PRL201+1
* PRL203: .set PRL202+1
* PRL204: .set PRL203+1
* PRL205: .set PRL204+1
* PRL2
```



```

BDATA: .set
TRANS: .set
COUNTER: .set
BIFSK: .set
AIFSK: .set
A2FSK: .set
FADDR: .set
FSKFLG: .set
ORLNG: .set
DZONE: .set
LPOUT: .set
CCITT: .set

; CURRENT BIRD FSK DATA (1111)
; = 20 USED IN TIMING DECISION
; TRANSITION COUNTER FOR RX TIMING
; COEF B1 OF PHASE ADJUST FIR
; COEF A1 OF FSK DEMOD FILTER
; COEF A2 OF FSK DEMOD FILTER
; ADDRESS OF 0 FREQUENCY
; ADDRESS OF 1 FREQUENCY
; FLAG TO INDICATE FSK OPERATION
; ORIGINATE/ANSWER MODE FLAG
; DEMO ZONE OF SLICER

```

```

*****
; ANG RPT
*****

```

```

*****
; ALPHA: .set
; ALPHA+1: .set
; ANSR+1: .set
; DM+1: .set
*****

```

```

*****
; BAUD COUNTER
*****

```

```

*****
; BCONTR: .set
; HYST+1
*****

```

```

*****
; PAGE 1 RAM ASSIGNMENTS
*****
;
; Z1: .set 0
; Z2: .set 12+1
; ST: .set 12+1
; STLSB: .set ST+1
; PRSSB: .set STLSB+1
; MEOSB: .set PRSSB+1
;

```

```

*****
;
; DIAGNOSTICS:
*****
;
; DTFEL: .set FRONT
; DTFPH: .set TIMING
;
; .text
; B START
*****
;
; COEFFICIENTS STORED IN PROGRAM ROM
*****
;
; PHASE ANGLE LOOK-UP TABLE
*****
;
; CONTAINS THE INCREMENT USED AS IDelta IN THE CARRIER GENERATION. THE
; TABLE GIVES THE ANGLES FOR 300 BPS, (V21, BELL 103) ORIGIN/ANSW MODES.
;
; NOTE: INCREMENTS ARE IN UNITS OF PI/128 TIMES 236 (UPPER 8 BITS OF DATA
; EVENTUALLY REPRESENT TABLE INDEX)
*****
;
; FSKTAB: .set $
; .word 0216dh ; FSK, 103, ORIGINATE, 1 1270 HZ
; .word 0168bh ; FSK, 103, ORIGINATE, 0 1070 HZ
; .word 01852h ; COEFF B1 FOR 2125 HZ FREQ (0.4293)
; .word 07fffh ; GAIN FOR FSK DEMOD LPF (1)
;
; .word 01a22h ; FSK, V-21, ORIGINATE, 1 1180 HZ
; .word 01f77h ; FSK, V-21, ORIGINATE, 0 980 HZ
; .word 2257ah ; COEFF B1 FOR 1750 HZ FREQ (0.43)
; .word 07fffh ; GAIN FOR FSK DEMOD LPF (0.5)
;
; .word 03a53h ; FSK, 103, ANSWER, 1 2225 HZ
; .word 03a00h ; FSK, 103, ANSWER, 0 2025 HZ
; .word 0a6ah ; COEFF B1 FOR 1170 HZ FREQ (0.3691)
; .word 0a9a3h ; GAIN FOR FSK DEMOD LPF (0.8223)
;
; .word 02000h ; FSK, V-21, ANSWER, 1 1850 HZ
; .word 03153h ; FSK, V-21, ANSWER, 0 1450 HZ
; .word 10747 ; COEFF B1 FOR 1080 HZ FREQ (0.3)
; .word 22057 ; GAIN FOR FSK DEMOD LPF (0.5)
;
; .word 03000h ; 2100 HZ ANSWER TONE
; .word 03a53h ; 2225 HZ ANSWER TONE
; .word 50 ; DEMO ZONE OF FSK DEMOD SLICER
; .word 0a999h ; COEF A1 OF FSK DEMOD FILTER
; .word 0a6ah ; COEF A2 OF FSK DEMOD FILTER
;

```

```
*****  
# COMMAND MODE SUBROUTINE LOOK-UP TABLE  
#  
# IN COMMAND MODE, EACH COMMAND BYTE CORRESPONDS TO AN ENTRY HERE WHICH  
# CONTAINS THE NAME OF THE APPROPRIATE SUBROUTINE TO CALL TO EXECUTE THE  
# COMMAND. THE NUMBER OF UNDEFINED SUBROUTINES (AT ADDRESSES 00, 01, 02,  
# 04, 06, 0A, 0B, 0C, AND 0EH) MEANS INCREASED SYSTEM EXPANSION IS EASILY  
# ACCOMMODATED.  
# *****  
#  
# CMTLTL: .set $  
#           : word NONE ; 00H NOT DEFINED  
#           : word NONE ; 1Ah NOT DEFINED  
#           : word NONE ; 2Ah NOT DEFINED  
#           : word RESET ; 3Ah RESET THE DSP  
#           : word NONE ; 4Ah NOT DEFINED  
#           : word FSXSET ; 5Ah FSK DATA MODE  
#           : word NONE ; 6Ah NOT DEFINED  
#           : word INMODE ; 7Ah RECVR. MODE SETUP  
#           : word INMODE ; 8Ah TRANS. MODE SETUP  
#           : word DTNFF ; 9Ah TRANSMIT DTNF CHARACTER  
#           : word NONE ; AAh NOT DEFINED  
#           : word NONE ; BAh NOT DEFINED  
#           : word OPER ; CAh SET OPERATING MODE  
#           : word NONE ; EAh NOT DEFINED  
#           : word PROTO ; FAh PROTOCOL SELECT  
  
*****  
# RECEIVER SUBROUTINE TABLE  
#  
# USED TO SELECT DISTRIBUTED RECEIVER TASKS. THIS IS IMPORTANT IN HIGHER  
# FUNCTIONALITY SYSTEMS THAT CANNOT IMPLEMENT TASKS IN A SINGLE SAMPLE  
# PERIOD.  
# *****  
#  
# RSEINTH: .set $  
#          : word NONE ;  
#          : word NONE ;  
#          : word NONE ;  
#          : word NONE ;  
#          : word NONE ;  
#          : word NONE ;  
#          : word NONE ;  
#          : word NONE ;  
#          : word RTSKLO ;  
#          : word NONE ;  
#          : word NONE ;
```

```

*****
* DTIME TONE TABLE:
*
* FIRST ENTRY REPRESENTS LOW FREQUENCY
* SECOND ENTRY REPRESENTS HIGH FREQUENCY
*
* DELTA = ( F / F ) * N
* S
*
* WITH N = 256 TABLE SIZE
* F = 9600 HZ
* S
* F = FREQUENCY OF INTEREST
*
*****
* DATA FORMAT IS 57.8 TO BE AS STEP SIZE. THE TABLE ENTRIES ARE HOWEVER,
* TREATED AS 16 BIT UNSIGNED INTEGERS. A MULTIPLICATION OF DELTA BY 256
* DOES THE NECESSARY CONVERSION IN FORMATS.
*
*****
*
* TONTBL: .word 01918h ; 0 LOW FREQ
* .word 02340h ; LOW FREQ GAIN
* .word 02640h ; HI FREQ GAIN
*
* .word 01296h ; 1
* .word 0203Eh ; LOW FREQ GAIN
* .word 0800h ; HI FREQ GAIN
*
* .word 01296h ; 2
* .word 02340h ; LOW FREQ GAIN
* .word 0800h ; HI FREQ GAIN
*
* .word 01296h ; 3
* .word 02762h ; LOW FREQ GAIN
* .word 0840h ; HI FREQ GAIN
*
* .word 01488h ; 4
* .word 0203Eh ; LOW FREQ GAIN
* .word 02E0h ; HI FREQ GAIN
*
* .word 01488h ; 5
* .word 02340h ; LOW FREQ GAIN
* .word 0420h ; HI FREQ GAIN
*
* .word 01488h ; 6

```

```

.word 02762h ; LOW FREQ GAIN
.word 0340h ; HI FREQ GAIN
*
* .word 01488h ; 7
* .word 0203Eh ; LOW FREQ GAIN
* .word 0240h ; HI FREQ GAIN
*
* .word 01488h ; 8
* .word 02340h ; LOW FREQ GAIN
* .word 0240h ; HI FREQ GAIN
*
* .word 01488h ; 9
* .word 02762h ; LOW FREQ GAIN
* .word 0290h ; HI FREQ GAIN
*
* .word 01296h ; A
* .word 028ECh ; LOW FREQ GAIN
* .word 0492h ; HI FREQ GAIN
*
* .word 01488h ; B
* .word 028ECh ; LOW FREQ GAIN
* .word 0492h ; HI FREQ GAIN
*
* .word 01488h ; C
* .word 028ECh ; LOW FREQ GAIN
* .word 0492h ; HI FREQ GAIN
*
* .word 01918h ; D
* .word 028ECh ; LOW FREQ GAIN
* .word 0492h ; HI FREQ GAIN
*
* .word 01918h ; E (e)
* .word 0203Eh ; LOW FREQ GAIN
* .word 0300h ; HI FREQ GAIN
*
* .word 01918h ; F (f)
* .word 02762h ; LOW FREQ GAIN
* .word 0300h ; HI FREQ GAIN
*

```

```

*****
*
* ADDITIONAL TABLES
*
*****
*
* .copy "COSINL.M00" ; COSINE FUNCTION TABLE
*
*****
*
* COSINE LOOKUP TABLE:
*
* 257 ENTRIES OVER THE RANGE [0,P1], THE RESOLUTION OF THE TABLE IS
* THEREFORE:
*
* (180 / 256 ) = 0.703125 DEGREES
*
*****
*
* COSOFF: .set $
*
* COSINE TABLE LENGTH = 512
*
* ANGLE = 0.0000 COSINE = 1.000000
* ANGLE = 0.7031 COSINE = 0.999925
* ANGLE = 1.4063 COSINE = 0.999699
* ANGLE = 2.1094 COSINE = 0.999322
* ANGLE = 2.8125 COSINE = 0.998795
* ANGLE = 3.5156 COSINE = 0.998118
* ANGLE = 4.2188 COSINE = 0.997291
* ANGLE = 4.9219 COSINE = 0.996313
* ANGLE = 5.6250 COSINE = 0.995185
* ANGLE = 6.3281 COSINE = 0.993907
* ANGLE = 7.0313 COSINE = 0.992480
* ANGLE = 7.7344 COSINE = 0.990903
* ANGLE = 8.4375 COSINE = 0.989177
* ANGLE = 9.1406 COSINE = 0.987301
* ANGLE = 9.8438 COSINE = 0.985278
* ANGLE = 10.5469 COSINE = 0.983104
* ANGLE = 11.2500 COSINE = 0.980785
* ANGLE = 11.9531 COSINE = 0.978317
* ANGLE = 12.6563 COSINE = 0.975702
* ANGLE = 13.3594 COSINE = 0.972940
* ANGLE = 14.0625 COSINE = 0.969931
* ANGLE = 14.7656 COSINE = 0.966676
* ANGLE = 15.4688 COSINE = 0.963176
* ANGLE = 16.1719 COSINE = 0.959431
* ANGLE = 16.8750 COSINE = 0.955440
* ANGLE = 17.5781 COSINE = 0.951206
* ANGLE = 18.2813 COSINE = 0.946828
* ANGLE = 18.9844 COSINE = 0.942307
* ANGLE = 19.6875 COSINE = 0.937544
* ANGLE = 20.3906 COSINE = 0.932539
* ANGLE = 21.0938 COSINE = 0.927293
* ANGLE = 21.7969 COSINE = 0.921806
* ANGLE = 22.5000 COSINE = 0.916080
* ANGLE = 23.2031 COSINE = 0.911114
*
*****

```

```

.word 07305h
.word 07405h
.word 07505h
.word 07605h
.word 07705h
.word 07805h
.word 07905h
.word 07A05h
.word 07B05h
.word 07C05h
.word 07D05h
.word 07E05h
.word 07F05h
.word 08005h
.word 08105h
.word 08205h
.word 08305h
.word 08405h
.word 08505h
.word 08605h
.word 08705h
.word 08805h
.word 08905h
.word 08A05h
.word 08B05h
.word 08C05h
.word 08D05h
.word 08E05h
.word 08F05h
.word 09005h
.word 09105h
.word 09205h
.word 09305h
.word 09405h
.word 09505h
.word 09605h
.word 09705h
.word 09805h
.word 09905h
.word 09A05h
.word 09B05h
.word 09C05h
.word 09D05h
.word 09E05h
.word 09F05h
.word 0A005h
.word 0A105h
.word 0A205h
.word 0A305h
.word 0A405h
.word 0A505h
.word 0A605h
.word 0A705h
.word 0A805h
.word 0A905h
.word 0AA05h
.word 0AB05h
.word 0AC05h
.word 0AD05h
.word 0AE05h
.word 0AF05h
.word 0B005h
.word 0B105h
.word 0B205h
.word 0B305h
.word 0B405h
.word 0B505h
.word 0B605h
.word 0B705h
.word 0B805h
.word 0B905h
.word 0BA05h
.word 0BB05h
.word 0BC05h
.word 0BD05h
.word 0BE05h
.word 0BF05h
.word 0C005h
.word 0C105h
.word 0C205h
.word 0C305h
.word 0C405h
.word 0C505h
.word 0C605h
.word 0C705h
.word 0C805h
.word 0C905h
.word 0CA05h
.word 0CB05h
.word 0CC05h
.word 0CD05h
.word 0CE05h
.word 0CF05h
.word 0D005h
.word 0D105h
.word 0D205h
.word 0D305h
.word 0D405h
.word 0D505h
.word 0D605h
.word 0D705h
.word 0D805h
.word 0D905h
.word 0DA05h
.word 0DB05h
.word 0DC05h
.word 0DD05h
.word 0DE05h
.word 0DF05h
.word 0E005h
.word 0E105h
.word 0E205h
.word 0E305h
.word 0E405h
.word 0E505h
.word 0E605h
.word 0E705h
.word 0E805h
.word 0E905h
.word 0EA05h
.word 0EB05h
.word 0EC05h
.word 0ED05h
.word 0EE05h
.word 0EF05h
.word 0F005h
.word 0F105h
.word 0F205h
.word 0F305h
.word 0F405h
.word 0F505h
.word 0F605h
.word 0F705h
.word 0F805h
.word 0F905h
.word 0FA05h
.word 0FB05h
.word 0FC05h
.word 0FD05h
.word 0FE05h
.word 0FF05h

```

```

: ANGLE = 23.9063 COSINE = 0.914210
: ANGLE = 24.6094 COSINE = 0.909168
: ANGLE = 25.3125 COSINE = 0.903989
: ANGLE = 26.0156 COSINE = 0.898675
: ANGLE = 26.7188 COSINE = 0.893224
: ANGLE = 27.4219 COSINE = 0.887640
: ANGLE = 28.1250 COSINE = 0.881921
: ANGLE = 28.8281 COSINE = 0.876070
: ANGLE = 29.5313 COSINE = 0.870087
: ANGLE = 30.2344 COSINE = 0.863973
: ANGLE = 30.9375 COSINE = 0.857729
: ANGLE = 31.6406 COSINE = 0.851355
: ANGLE = 32.3438 COSINE = 0.844854
: ANGLE = 33.0469 COSINE = 0.838225
: ANGLE = 33.7500 COSINE = 0.831470
: ANGLE = 34.4531 COSINE = 0.824589
: ANGLE = 35.1563 COSINE = 0.817585
: ANGLE = 35.8594 COSINE = 0.810457
: ANGLE = 36.5625 COSINE = 0.803208
: ANGLE = 37.2656 COSINE = 0.795837
: ANGLE = 37.9688 COSINE = 0.788347
: ANGLE = 38.6719 COSINE = 0.780737
: ANGLE = 39.3750 COSINE = 0.773011
: ANGLE = 40.0781 COSINE = 0.765168
: ANGLE = 40.7813 COSINE = 0.757209
: ANGLE = 41.4844 COSINE = 0.749137
: ANGLE = 42.1875 COSINE = 0.740951
: ANGLE = 42.8906 COSINE = 0.732655
: ANGLE = 43.5938 COSINE = 0.724247
: ANGLE = 44.2969 COSINE = 0.715731
: ANGLE = 45.0000 COSINE = 0.707107
: ANGLE = 45.7031 COSINE = 0.698377
: ANGLE = 46.4063 COSINE = 0.689541
: ANGLE = 47.1094 COSINE = 0.680601
: ANGLE = 47.8125 COSINE = 0.671559
: ANGLE = 48.5156 COSINE = 0.662416
: ANGLE = 49.2188 COSINE = 0.653173
: ANGLE = 49.9219 COSINE = 0.643822
: ANGLE = 50.6250 COSINE = 0.634374
: ANGLE = 51.3281 COSINE = 0.624860
: ANGLE = 52.0313 COSINE = 0.615232
: ANGLE = 52.7344 COSINE = 0.605512
: ANGLE = 53.4375 COSINE = 0.595700
: ANGLE = 54.1406 COSINE = 0.585799
: ANGLE = 54.8438 COSINE = 0.575809
: ANGLE = 55.5469 COSINE = 0.565733
: ANGLE = 56.2500 COSINE = 0.555571
: ANGLE = 56.9531 COSINE = 0.545326
: ANGLE = 57.6563 COSINE = 0.534998
: ANGLE = 58.3594 COSINE = 0.524590
: ANGLE = 59.0625 COSINE = 0.514103
: ANGLE = 59.7656 COSINE = 0.503539
: ANGLE = 60.4688 COSINE = 0.492899
: ANGLE = 61.1719 COSINE = 0.482184

```

08C57h .word ANGLE = 61.8750 COSINE = 0.471397
 08C63h .word ANGLE = 62.5781 COSINE = 0.446039
 08C69h .word ANGLE = 63.2813 COSINE = 0.419612
 08C6Eh .word ANGLE = 63.9844 COSINE = 0.393117
 08C73h .word ANGLE = 64.6875 COSINE = 0.367556
 08C79h .word ANGLE = 65.3906 COSINE = 0.342430
 08C7Eh .word ANGLE = 66.0938 COSINE = 0.317842
 08C83h .word ANGLE = 66.7969 COSINE = 0.293792
 08C89h .word ANGLE = 67.5000 COSINE = 0.269384
 08C8Eh .word ANGLE = 68.2031 COSINE = 0.245619
 08C93h .word ANGLE = 68.9063 COSINE = 0.222500
 08C99h .word ANGLE = 69.6094 COSINE = 0.199919
 08CA6h .word ANGLE = 70.3125 COSINE = 0.177870
 08CB3h .word ANGLE = 71.0156 COSINE = 0.156363
 08CB9h .word ANGLE = 71.7188 COSINE = 0.135397
 08CC6h .word ANGLE = 72.4219 COSINE = 0.114970
 08CCBh .word ANGLE = 73.1250 COSINE = 0.095182
 08CD0h .word ANGLE = 73.8281 COSINE = 0.075932
 08CD6h .word ANGLE = 74.5313 COSINE = 0.057230
 08CE3h .word ANGLE = 75.2344 COSINE = 0.039076
 08CE9h .word ANGLE = 75.9375 COSINE = 0.021470
 08CEEh .word ANGLE = 76.6406 COSINE = 0.004412
 08CF3h .word ANGLE = 77.3438 COSINE = 0.000000
 08CF9h .word ANGLE = 78.0469 COSINE = 0.000000
 08CDEh .word ANGLE = 78.7500 COSINE = 0.000000
 08CE3h .word ANGLE = 79.4531 COSINE = 0.000000
 08CE9h .word ANGLE = 80.1563 COSINE = 0.000000
 08CEEh .word ANGLE = 80.8594 COSINE = 0.000000
 08CF3h .word ANGLE = 81.5625 COSINE = 0.000000
 08CF9h .word ANGLE = 82.2656 COSINE = 0.000000
 08CDEh .word ANGLE = 82.9688 COSINE = 0.000000
 08CE3h .word ANGLE = 83.6719 COSINE = 0.000000
 08CE9h .word ANGLE = 84.3750 COSINE = 0.000000
 08CEEh .word ANGLE = 85.0781 COSINE = 0.000000
 08CF3h .word ANGLE = 85.7813 COSINE = 0.000000
 08CF9h .word ANGLE = 86.4844 COSINE = 0.000000
 08CDEh .word ANGLE = 87.1875 COSINE = 0.000000
 08CE3h .word ANGLE = 87.8906 COSINE = 0.000000
 08CE9h .word ANGLE = 88.5938 COSINE = 0.000000
 08CEEh .word ANGLE = 89.2969 COSINE = 0.000000
 08CF3h .word ANGLE = 90.0000 COSINE = 0.000000
 08CF9h .word ANGLE = 90.7031 COSINE = -0.012272
 08CDEh .word ANGLE = 91.4063 COSINE = -0.024542
 08CE3h .word ANGLE = 92.1094 COSINE = -0.036808
 08CE9h .word ANGLE = 92.8125 COSINE = -0.049069
 08CEEh .word ANGLE = 93.5156 COSINE = -0.061322
 08CF3h .word ANGLE = 94.2188 COSINE = -0.073566
 08CF9h .word ANGLE = 94.9219 COSINE = -0.085798
 08CDEh .word ANGLE = 95.6250 COSINE = -0.098018
 08CE3h .word ANGLE = 96.3281 COSINE = -0.110223
 08CE9h .word ANGLE = 97.0313 COSINE = -0.122412
 08CEEh .word ANGLE = 97.7344 COSINE = -0.134582
 08CF3h .word ANGLE = 98.4375 COSINE = -0.146722

08E00h .word ANGLE = 99.1406 COSINE = -0.158859
 08E06h .word ANGLE = 99.8438 COSINE = -0.170963
 08E0Bh .word ANGLE = 100.5469 COSINE = -0.183041
 08E10h .word ANGLE = 101.2500 COSINE = -0.195092
 08E16h .word ANGLE = 101.9531 COSINE = -0.207113
 08E1Bh .word ANGLE = 102.6563 COSINE = -0.219103
 08E20h .word ANGLE = 103.3594 COSINE = -0.231060
 08E26h .word ANGLE = 104.0625 COSINE = -0.242982
 08E2Bh .word ANGLE = 104.7656 COSINE = -0.254867
 08E30h .word ANGLE = 105.4688 COSINE = -0.266714
 08E36h .word ANGLE = 106.1719 COSINE = -0.278521
 08E3Bh .word ANGLE = 106.8750 COSINE = -0.290286
 08E40h .word ANGLE = 107.5781 COSINE = -0.302008
 08E46h .word ANGLE = 108.2813 COSINE = -0.313683
 08E4Bh .word ANGLE = 108.9844 COSINE = -0.325312
 08E50h .word ANGLE = 109.6875 COSINE = -0.336892
 08E56h .word ANGLE = 110.3906 COSINE = -0.348420
 08E5Bh .word ANGLE = 111.0938 COSINE = -0.359897
 08E60h .word ANGLE = 111.7969 COSINE = -0.371319
 08E66h .word ANGLE = 12.5000 COSINE = -0.382685
 08E6Bh .word ANGLE = 13.2031 COSINE = -0.393994
 08E70h .word ANGLE = 13.9063 COSINE = -0.405243
 08E76h .word ANGLE = 14.6094 COSINE = -0.416431
 08E7Bh .word ANGLE = 15.3125 COSINE = -0.427557
 08E80h .word ANGLE = 16.0156 COSINE = -0.438618
 08E86h .word ANGLE = 16.7188 COSINE = -0.449613
 08E8Bh .word ANGLE = 17.4219 COSINE = -0.460541
 08E90h .word ANGLE = 18.1250 COSINE = -0.471399
 08E96h .word ANGLE = 18.8281 COSINE = -0.482186
 08E9Bh .word ANGLE = 19.5313 COSINE = -0.492900
 08EE0h .word ANGLE = 20.2344 COSINE = -0.503540
 08EE6h .word ANGLE = 20.9375 COSINE = -0.514105
 08EEBh .word ANGLE = 21.6406 COSINE = -0.524592
 08EF0h .word ANGLE = 22.3438 COSINE = -0.535000
 08EF6h .word ANGLE = 23.0469 COSINE = -0.545327
 08EFBh .word ANGLE = 23.7500 COSINE = -0.555572
 08F00h .word ANGLE = 24.4531 COSINE = -0.565734
 08F06h .word ANGLE = 25.1563 COSINE = -0.575810
 08F0Bh .word ANGLE = 25.8594 COSINE = -0.585800
 08F10h .word ANGLE = 26.5625 COSINE = -0.595701
 08F16h .word ANGLE = 27.2656 COSINE = -0.605513
 08F1Bh .word ANGLE = 27.9688 COSINE = -0.615224
 08F20h .word ANGLE = 28.6719 COSINE = -0.624842
 08F26h .word ANGLE = 29.3750 COSINE = -0.634395
 08F2Bh .word ANGLE = 30.0781 COSINE = -0.643894
 08F30h .word ANGLE = 30.7813 COSINE = -0.653317
 08F36h .word ANGLE = 31.4844 COSINE = -0.662618
 08F3Bh .word ANGLE = 32.1875 COSINE = -0.671751
 08F40h .word ANGLE = 32.8906 COSINE = -0.680803
 08F46h .word ANGLE = 33.5938 COSINE = -0.689778
 08F4Bh .word ANGLE = 34.2969 COSINE = -0.698678
 08F50h .word ANGLE = 35.0000 COSINE = -0.707499
 08F56h .word ANGLE = 35.7031 COSINE = -0.716233
 08F5Bh .word ANGLE = 36.4063 COSINE = -0.724889

SINE0:


```

*****
*
*   INITIALIZE PAGE 0 DATA
*
*   SYSTEM IS ORIGINALLY INITIALIZED AT A PSEUDO 1200 BPS, TX, RX IDLE LINE
*   MODE. 16 SAMPLES/BAUD TO ACCOMMODATE THE START-UP CONDITION OF THE
*   TMS70C2400 MODEN CONTROLLER.
*
*****
*
*   LBRX      0
*   LOCK      8
*   SACL      INOUT
*   LOCK      2
*   SACL      STATUS
*   LOCK      15
*   SACL      SCLT
*   SACL      XSCNTR
*   SACL      RSCNTR
*   LOCK      1
*   SACL      ONE
*
*****
*
*   INITIALIZE SERIAL PORT CONTROL REGISTERS
*
*****
*
*   DRL
*   LOCK      TXPR
*   OUT      TXPR,PA0
*   LOCK      DRL
*   TXLR      TXPR
*   OUT      TXPR,PA1
*   LOCK      DRL
*   TXLR      TXPR
*   OUT      TXPR,PA0
*
*****
*
*   INITIALIZE ABC GAIN AND LOCK INDICATOR. (MOD. 5/79)
*
*****
*
*   LOCK      MAINLP
*   TXLR      ALPHA      ; SET ALPHA TO ITS MAX POSSIBLE VALUE
*
*****
*
*   INITIALIZE STATUS WORD TO SET AFE GAIN STAGE ON
*
*****
*
*   LOCK      000h

```

```

*****
*
*   SACL      STMRD
*
*****
*
*   INITIALIZE HYSTERESIS COUNTER TO 0000h
*
*****
*
*   LAC      ONE,15
*   SACL      HYST
*
*****
*
*   MAIN PROGRAM SEQUENCER
*
*   THE MAIN PROGRAM SEQUENCER PROVIDES THE TIMING FOR THE MAIN PROGRAM LOOP
*   AND CALLS THE VARIOUS SUBROUTINES AT THE APPROPRIATE TIMES. THE MAIN LOOP
*   IS COMPLETED ONCE EVERY BAUD INTERVAL OR EQUIVALENTLY 300 TIMES/SEC FOR
*   FSK.
*
*   THE PCA CODEC HAS A FIXED SAMPLING RATE OF 9.6 KHZ, WHICH MEANS THAT IN
*   THE FSK MODE THE DSP BAUD PERIOD CORRESPONDS TO 32 PCA SAMPLES.
*
*****
*
*   WAIT      IN      TXPR,PA0      ; WAIT FOR FR INTERRUPT FLAG
*   LOCK      8                    ; IF NOT LOOP HERE
*   AND      TXPR      ; IF YES TRANSMIT AND RECEIVE FROM PORT 1
*   BZ      WAIT      ; AND RESET THE INTERRUPT FLAG
*
*   SEND      OUT      XOUT,PA1
*   IN      RXIN,PA1
*   LOCK      DRL
*   TXLR      TXPR
*   OUT      TXPR,PA0
*
*****
*
*   EXECUTE TRANSMITTER TASK FOR TIME SLOT AND UPDATE SAMPLE COUNTER.
*
*****
*
*   LAC      XSCNTR
*   SUB      SCLT      ; IF ZERO CALL CCI
*   BZ      SEND01
*
*   SEND01    CALL      CCI      ; CALL COMMAND INTERPRETER
*
*****
*
*   EXECUTE THE SAMPLE TASKS
*
*****
*

```

```

*****
*
*   EXECUTE RECEIVER TASK FOR TIME SLOT AND UPDATE RECEIVER SAMPLE COUNTER.
*   IF RECEIVER IN IDLE MODE RETURN TO WAIT STATE
*
*****
*
*   LACK
*   AND STATUS
*   INZ SEQ4
*   LAC RSCNTR
*   SUB ONE
*   BGEZ SEQ42
*   LAC SONT
*   OUT STWRD,PHS
*   RSCNTR SACL
*   B WAIT
*
*****
*
*   USE ROUTINE 'RTSK' TO PERFORM FSK DEMODULATION
*
*****
*
*   CALL RTSK
*   LACK 3
*   AND STATUS
*   BZ DECRS
*
*   ; CHECK FSK OPERATION
*   ; IF SQ, JUST DECREMENT RX SAMPLE COUNTER
*
*   LACK RSE0TB
*   AND RSCNTR
*   TLRL TRP1
*   LAC TRP1
*   CALA
*
*   ; IN HANDSHAKING MODE CALL RTSK
*   ; SUBROUTINE ONCE PER SAMPLE. ONLY ONE
*   ; NON-TRIVIAL FUNCTION, RTSK10, IS
*   ; ACTUALLY CALLED.
*
*   DECRS LAC RSCNTR
*   SUB ONE
*   BGEZ SEQ46
*   CALL AOC
*
*   ; IF NOT THE END OF BAND, JUST CONTINUE
*   ; ELSE DO AOC ONCE PER BAND.
*
*   SEQ46 LAC SONT
*   SEQ46 RSCNTR
*   B WAIT
*
*****
*
*   SUBROUTINES
*
*****
*
*   FSKT1: .set $
*           ZALL TOPHLE
*           CALL SIMLEN
*           ; BRING IN TI PHOLE
*           ; GENERATE TIME AT APPROPRIATE FREQ
*
*****

```



```
*
*      DTL6AG          ; RESULT RETURNED IN TYP3
*      ISDTNF
*
*
*      LAC              ; AS12 FORMAT
*      BKZ
*
*      LAC              ; AS12 FORMAT
*      SACH             ; 4S12 FORMAT
*      LT              ; 4S12 FORMAT
*      XOUT             ; 4S12 FORMAT
*      OT00h           ; 4S12 FORMAT
*      PKC              ; 4S12 FORMAT
*      SACH             ; 4S12 FORMAT
*      B               ; 4S12 FORMAT
*
*      ISDTNF          $
*      LAC              ; 2S30 FORMAT - LOWER FREQUENCY
*      TYP3_15         ; 2S30 FORMAT - LOWER FREQUENCY
*      SACH            ; 2S14 FORMAT - LOWER FREQUENCY
*      TYP3
*
*      ZALS             ; IN DTNF MODE, HIGH FREQ IS HANDLED BY
*                        ; BY RPHS
*      RPHS             ; GENERATE HIGH FREQUENCY TONE
*
*      CALL             ; INCORPORATE HIGH FREQUENCY ANGLE
*      SINGEN           ; INCORPORATE HIGH FREQUENCY PHASE ANGLE
*      ZALS             ; BY SECOND TONE FREQUENCY
*      ADUS             ; STORE AWAY
*      RPTFRQ           ;
*      SACL             ;
*      RPHS             ;
*
*
*      LAC              ; TYP3_15
*      SACH             ; TYP3_15
*      LT              ; TYP3
*      TYP3            ; TYP3
*      RPY              ; DTNFI
*      PAC              ; DTNFI
*      LT              ; TYP3
*      RPY              ; DTNFI
*      APAC             ; DTNFI
*      SACH             ; TYP3
*      XOUT_4           ; TYP3
*
*      .set            $
*      ZALS             ; TYP3
*      ADUS             ; TYP3
*      SACL             ; TYP3
*
*      NONE            ; RET
*
*      .copy            ; "CCIDTLM_A00" ; INCLUDES CODE FOR DTNF
*
*
*      *****
*
*      CONTROLLER COMMAND INTERPRETER (CCI)
*
*      THE FOLLOWING CODE READS A COMMAND FROM THE TMS702400 ON PORT 5 AND
*      INTERPRETS IT ACCORDING TO THE RULES SPECIFIED IN THE CONTROLLER-OSP
*      INTERFACE DOCUMENT. THE 320 READS ONE COMMAND EVERY 8000 PERIOD. THE BAD
*      RATE IS INITIALLY SET TO 600, AND THE BAD CLOCK IS DERIVED FROM THE
*      SERIAL PORT FR SIGNAL.
*
*****
```

```

SUB      ONE.2
BLEZ    CC11      ; IF COMMAND LARGER THAN 32 EXIT COMMAND
RET

*****
*
* CALL THE APPROPRIATE SERVICE SUBROUTINE (REFER TO CH01B1 TABLE).
*
*****
CC11     LACK      CH01B1      ; BASE OF COMMAND TABLE
        AND      TMP2      ; AND COMMAND PROCE
        TBLR     TMP2      ; READ ADDRESS FROM TABLE
        LAC      TMP2      ; LOAD SUB. ADDR. INTO ACC.
        CALA     ; CALL SERVICE SUBROUTINE
        RET      ; EXIT COMMAND INTERPRETER.
*****
*
* COMMAND INTERPRETER SUBROUTINES
*
*****
*
* PROTOCOL SELECT COMMAND
*
*****
PROTO:   .set     $
*****
*
* EXTRACT TWO LSB'S OF COMMAND
*
*****
LACK     03h      ; MASK OFF BITS 2 AND 3 OF COMMAND
AND      XDATA
*****
*
* SET SPEED BITS STATUS REGISTER:
*
*****
BITS 1 AND 0 = 00 FOR 300 BPS (FSK)
           = 01 RESERVED
           = 10 RESERVED
           = 11 RESERVED
*****
*
*
*****
SACL     TMP1
LACK     0Fh
AND      STATUS   ; ZERO BITS 0 AND 1
SACL     STATUS

```

```

*****
*
* DETERMINE FSK FREQUENCIES AND SET BAUD COUNTER, ALSO SET OTHER FSK
* SIGNAL PROCESSING PARAMETERS
*
*****
PROTO2   LACK     31      ; BAUD COUNTER IS 32
        SACL     SCNT
*****
*
*
        LAC      ONE.2
        AND      XDATA
        SACL     CC11T
*****
*
*
        LAC      ONE.3
        SACL     FSKFGL
*****
*
*****
*
* ACCUMULATOR NOW CONTAINS THE NUMERAL 8 LOGICAL AND HENCE IDENTIFIES
* ORIGINATE./ANSWER MODES
*
*****
*
        AND      XDATA      ; ISOLATE ORIGINATE./ANSWER BIT
        SACL     OPLAG      ; SET OPLAG 0 0 IN ANSWER MODE = 0 IN
                             ; ORIGINATE MODE
        LACK     02h      ; MASK TWO LSB'S OF COMMAND
        AND      XDATA
        SACL     TMP1
        LACK     FSKTBL
        AND      TMP1      ; ADD BASE OF FSK TABLE
        SACL     FIADD      ; 1 FREQUENCY
        ADD      ONE
        SACL     FIADD      ; 0 FREQUENCY ADDRESS
        AND      ONE
        TBLR     BIFSK      ; FSK LOWPASS FILTER COEFFICIENT
        ADD      ONE
        TBLR     GAIN      ; FSK MODE GAIN
        SUB      ONE
        SACL     TMP1      ; TMP1 NOW POINTS TO BIFSK
*****
*
*****
*
* SET FSK TIMING RECOVERY PARAMETERS.
*
*****
*
        LACK     02h      ; ACC = 12
        SACL     TRANS      ; TRANS = 12
*****
*
*****
*
* SET FSK RECEIVE FILTER COEFFICIENTS AND SLICER DEAD ZONE
*
*****

```

```

*****
# SELECT GUARD TONE
#
# GUARD RET
#
*****
#
# SET OPERATING MODE
#
*****
#
# OPER LACK LWRK ; ZERO ANALOG LOOPBACK
# AND STATUS ; LOC. DIG. LOOPBACK BITS
#
*****
#
# CHECK OPERATING MODE
#
*****
#
# LACK G3H ; MASK OFF BITS 2 AND 3 OF COMMAND
# AND XD0A
# BL OPER1 ; IF ZERO => LINE MODE (RET)
# SUB ONE
# BL ANLB ; IF ONE => ANALOG LOOPBACK
#
# OPER1 RET
#
# ANLB LACK ANWSK ; SET ANALOG LOOPBACK
# OR STATUS ; STATUS BIT
# SKCL STATUS
# RET
#
*****
#
# DIAL DTMF
#
# DTMF SET-UP ROUTINE : LOOKUP THE LOW AND HIGH FREQUENCIES CORRESPONDING
# TO EVERY DIGIT AND PLACE IN TFRQ AND RUFRQ RESPECTIVELY.
#
*****
#
# DTMF LACK 15 ; MASK FOR ISOLATING THE DIGIT
# AND XD0A
# SKCL TFRQ ; STORE AWAY TEMPORARILY
# LACK TONTEL ; BRING IN BASE ADDRESS OF TONE TABLE
# AND TFRQ,2 ; LEFT SHIFT IS REQUIRED AS THERE ARE FOUR
# ; ENTRIES PER DIGIT
# TBLR TFRQ ; READ LOW FREQ INTO TFRQ
# ADD ONE ; INCREMENT POINTER TOWARDS HIGH FREQ
# TBLR RUFRQ ; READ HIGH FREQ INTO RUFRQ
# ADD ONE ; READ IN LO FREQ GAIN
# DTML DTML ; READ IN HI FREQ GAIN
# TBLR DTML
#
*****

```



```

*****
*
* FSK DATA MODE
*
* SET UP FSK TRANSMIT FREQ ACCORDING TO THE TX DATA
*
*****
*
* FSKSET LACK 8 ; CHECK THE TRANSMITTED BIT
* AND DATA
* BZ ; IF ZERO, DATA MUST BE 0
* LAC FREQ ; POINT ACC TO 1 FREQ
* TBLR TIFREQ ; SET TX FREQ TO APPROPRIATE 1 FREQ
* RET
*
* DATAO LAC FREQ ; POINT ACC TO 0 FREQ
* TBLR TIFREQ ; SET TX FREQ TO APPROPRIATE 0 FREQ
*
* MFSK RET
*
*****
*
* RESET AND EQUALIZER ENABLE ROUTINES
*
*****
*
* RESET LACK 0B11
* SACK STARD
* OUT STARD,PHS
*
* 8 START
*
*****
*
* END CONTROLLER COMMAND INTERPRETER SUBROUTINES
*
*****

```

```

*
* .COPY "SINEN.A00"
*
*****
*
* SUBROUTINE : SINEN
*
* PURPOSE : SINE GENERATION
*
* TASK : GIVEN A COSINE TABLE WITH 257 VALUES AND START ADDRESS COSOFF, AND
* GIVEN AN ANGLE INDEX IN THE ACCUMULATOR, DETERMINE THE SINE OF THE ANGLE.
*
* ENTRY CONDITION : THE ANGLE INDEX MUST BE IN THE LOWER ACCUMULATOR.
*
* EXIT CONDITION : THE SINE OF THE ANGLE IS RETURNED IN TEMPORARY LOCATION
* TMP3.
*
* DESCRIPTION : THE COSINE LOOKUP TABLE CONTAINS 257 VALUES WITH:
*
*  $\cos(0) = 1.0$  AND  $\cos(256) = -1.0$ 
*
* HENCE ANGLE INDEX 0 MAPS TO ANGLE 0 AND ANGLE INDEX 256 MAPS TO  $\pi$ . THE
* SINE VALUE IS GENERATED BY SUBTRACTING FROM THE ANGLE INDEX THE INDEX
* CORRESPONDING TO  $\pi/2$ , TAKING THE ABSOLUTE VALUE, AND HENCE FORMING AN
* ADDRESS INTO THE LOOKUP TABLE.
*
* NO OF CYCLES: 17
*
* NO OF STACK LEVELS USED: 1
*
* THE ANGLE INDEX IS THE LOWER ACCUMULATOR
*
* ANGLE INDEX HAS SIS.0 FORMAT, MUST SUBTRACT  $\pi/2$  VALUE WHICH LAYS AT THE
* MIDDLE OF THE TABLE AND HAS SS14.0 FORMAT AS VIEWED IN SIS.0 FORMAT
*
*****
*
* SINEN SUB ONE,14 ; SUBTRACT INDEX OF  $\pi/2$ 
* SACK TMP3 ; PUT AWAY TEMPORARILY
* TALK TMP3 ; PREPARE FOR ABSOLUTE VALUE
* ABS ; TAKE ABSOLUTE VALUE
* SACK TMP3 ; PUT AWAY BEFORE RIGHT SHIFT
*
*****
*
* THE VALUE STORED IN TMP3 HAS SIS.0 FORMAT -- ALBEIT A POSITIVE NUMBER
*
* A LEFT SHIFT OF 9 BITS CORRESPONDS TO SS24.0 FORMAT AND SAVING THE HIGH
* ACCUMULATOR HAS A SS8.0 FORMAT
*
*****
*
* LAC TMP3,9 ; ISOLATE 8 MSB'S IN HIGH ACC
* SACK TMP3 ; PUT AWAY THE 8 MSB'S TEMPORARILY
*
*****

```

```

*****
*
* THE NEXT THREE INSTRUCTIONS ELIMINATE ANY SIGN EXTENSION BITS THAT MIGHT
* HAVE PROPAGATED
*
*****
*
* LAC TWP3
* AND
* SACL TWP3
*
*****
*
* FORM THE FINAL LOOK-UP ADDRESS
*
* BRING IN THE ADDRESS OFFSET.
*
* THE BASE ADDRESS IS IN 888.0 FORMAT, WHILE THE INDEX IS ALSO IN 888.0
* FORMAT.
*
*****
*
* LACK COSOFF
* AND TWP3 ; FORM FINAL LOOK-UP ADDRESS
* TBLR TWP3 ; READ SINE VALUE INTO TWP3
*
* RET
*
*****
*
* FSK DEMODULATION FILES
*
*****
*
* .copy "RSTSE.A00"
*
*****
*
* DATE: 5-29-86
*
* SUBROUTINE: RSTSK
*
* INCLUDES FSK RECEIVER/TIMING RECOVERY
*
* PURPOSE: RECEIVER PER SAMPLE TASK
*
* TASK: THIS SUBROUTINE COMBINES SMALLER MODULES TO PERFORM THE SIGNAL
* PROCESSING FUNCTIONS THAT ARE REQUIRED ON A PER SAMPLE BASIS
* (9600 Hz).
*
* ENTRY CONDITION: THE RECEIVED S/M SAMPLE IS IN RAM LOCATION RIN.
*
*****

```

```

*****
*
* RSTSK: .set $
* LAC RIN ; INPUT 14-BIT S/M SAMPLE
* SACL TWP1 ; ARL INTO TWP1
*
*****
*
* USE HIGH PASS FILTER. MAKE SURE INCOMING SAMPLE HAS NO SHIFT.
*
*****
*
* LAC TWP1,0 ; ARL
*
*****
*
* HIGH PASS FILTER THE INPUT
*
*****
*
* RLP: .set $
* LUPK 1
* SACL X1
* ZALS STL58
* ANDH ST
* SUB ST,TAU
* ANDH X1
* SUB X1,TAU-1
* SUBH X2
* ADD X2,TAU-1
* SACL STL58
* SACH ST
* DMOV X1
* LUPK 0
* SACH TWP1
*
*****
*
* RLP LEAVES THE SAMPLE IN TWP1. MULTIPLY IT BY AGC GAIN ALPHA. THE OUTPUT
* FORMAT IS 84.11 REQUIRING SOME MANIPULATIONS
*
*****
*
* LT TWP1 ; MULTIPLY BY AGC WORD
* RPY ALPHA
* PAC
*
*****
*
* SHIFT ACCUMULATOR EIGHT 4 TIMES BEFORE STORING
*
*****
*
* SACL TWP0
* SACH TWP1
*
*****

```

```

LAC      TMP0,8
SACH     TMP0
LAC      ONE,8      ; MASK OFF ANY SIGN EXTENSION
SUB      ONE
AND      TMP0
AND      TMP0
SACL     TMP0,8
LAC      TMP1,8
AND      TMP0
AND      TMP1
SACL     TMP1

*****
*      IN BOTH TMP1 AND TMP2 UPDATE THE SIGNAL POWER ESTIMATE AVESSOR
*
*      AVESSOR = AVESSOR + TMP1*2
*
*      AVESSOR IS ZEROED BY THE AGC ROUTINE ONCE PER BAUD.
*****

LAC      TMP1,15
SACH     TMP0
LT       TMP0
MPY      TMP1
PAC      AVESSOR
ANDH     AVESSOR
SACH     AVESSOR

*****
*      INCREASE SIGNAL ENERGY BY A FACTOR OF 4 FOR COMPATIBILITY WITH THE REST
*      OF THE RECEIVER
*****

LAC      TMP1,1
SACL     TMP1
SACL     TMP2

*****
*      CHECK FOR FSK OPERATION
*
*      LACK      3
*      AND      STATUS
*
*      LAC      TMP1,0
*      SACL     AVESSOR      ; PLACE RECEIVE SIGNAL IN AVESSOR
*      CALL     RUFASK      ; CALL FSK RECEIVING/TIMING RECOVERY
*      RET      CCITT,600
*      .copy
*****

*      CHECK FOR FSK OPERATION
*
*****
LAC      3
AND      STATUS
LT       TMP1
MPY      ONE,21      ; ASSUME V,21
; MULTIPLY BY 2.5
LAC      CCITT
BRL      V2LONG
MPYK     ONE,8h      ; MULTIPLY BY 3.0
PAC      V2LONG
SACL     TMP0
SACH     TMP1
LAC      TMP0,13
SACH     TMP0
LAC      ONE,13
SUB      ONE
AND      TMP0
SACL     TMP0
LAC      TMP1,13
AND      TMP0
SACL     AVESSOR
CALL     RUFASK
RET      .copy      *FREQ2,600"
*****
*      DEMODULATOR SECTION
*
*      THIS DESIGN IMPLEMENTS A DELAY OF 51P1/2
*
*      AGC
*
*      FSK DEMODULATOR
*
*      MEMORY CONFIGURATION (CONSECUTIVE ADDRESSES)
*      AVESSOR
*      PDEL0
*      PDEL1
*      PDEL2
*      LPDEL0
*      LPDEL1
*      LPDEL2
*****
RUFASK   SACH      ; SET OVERFLOW MODE
*****
*      TAKE PRODUCT FOR PRODUCT DEMOD SCHEME. ASSUME ANSWER MODE (2 SAMPLES)
*****

```



```

*
* RET
*
*****
*
* AGC FILES
*
*****
*
* .COPY "AGC.A00"
*
*****
*
* AGC.ASH
*
*****
*
* FRONT END AGC FUNCTION.
*
*****
*
* THIS AGC WAS REDESIGNED TO INCORPORATE THE FREEZE OF EQUALIZATION 5/29/87
*
*****
*
* THE AVERAGE SIGNAL SQUARED IS COMPUTED BY THE MAIN PROGRAM AND STORED IN
* AVESSOR, WHICH IS CLEARED BY THIS ROUTINE AFTER AVESSOR IS USED. THE
* ROUTINE USES A WINDOW WHOSE WIDTH DEPENDS ON THE MODULATION (1200, 2400)
* AND AN ERROR WEIGHTING WHICH ALSO DEPENDS ON THAT RATE. WE FIRST SET
* THOSE VALUES:
*
*****
*
* AGC
*
*****
*
* SUMM LAC ON : CHECK FOR AFE SWITCHING
*
*****
*
* BNL SWITCH
*
*****
*
* CHECK IF 2400 AND CHANGE THOSE VALUES
*
*****
*
* LACK 3
* AND STATUS : IF STATUS BITS 0 AND 1 > 2 => 2400
* SUB ONE.1
* BLEZ AGC0 : IF <= 2, DO NOT MODIFY THP0 AND THP1
*
*****
*
* FOR 2400, 2 -> THP0 AND 1200 -> THP1 -
*
*****
*
* LACK 2
* SACL THP0 : IT IS 2400
* LT ONE
* MPYK 1200
* PAC
* SACL THP1
* B AGC1

```

```

*
* AGC0
*
*****
*
* LACK 1
* SACL THP0 : IT IS 1200
* LT ONE : MEIGHTING FACTOR -> THP0
* MPYK 900
* PAC
* SACL THP1 : WINDOW -> THP1
*
*****
*
* NOW SUBTRACT REFERENCE FROM BAUD ENERGY TO GET ERROR. THE BAUD ENERGY IS
* IN SIO.5 FORMAT. THE AGC MAINTAINS THAT LEVEL AT 2.84+16 = 46.7 (H'386)
* IN SIO.5). THE AGCREF IS THEREFORE H'386
*
*****
*
* AGC1
*
*****
*
* LAC AVESSOR
* BEZ CONT1 : FOR NEGATIVE ENERGY SET TO MIN POSITIVE
* LAC ONE.15 : ENERGY LEVEL - FORCED SLEN MODE
* SUB ONE
* LT ONE
* MPYK AGCREF : AGCREF = H'386
* SPAC : AVESSOR - AGCREF -> AGC
*
*****
*
* COMPARE THE ERROR TO WINDOW (THP1).
*
*****
*
* IF ERROR > WINDOW => ERROR - WINDOW -> ERROR
* IF -WINDOW < ERROR < WINDOW => 0 -> ERROR
* IF ERROR < -WINDOW => THP0 + (ERROR + WINDOW) -> ERROR
*
*****
*
* IF THE AVERAGE BAUD ENERGY IS A, THE PEAK BAUD ENERGY FOR QAM SIGNALS IS
* 1.8 A AND THE MINIMUM IS 0.2 A. THE WINDOW IS THEREFORE CHOSEN TO BE 0.8
* A IN EITHER DIRECTION. WITH AGCREF = H'386, THE WINDOW IS H'492. FOR QPSK
* SIGNALS, THE VARIATIONS IN BAUD ENERGY ARE ENTIRELY DUE TO ISI AND
* DISTORTION AND THEREFORE THE WINDOW IS MUCH SMALLER (H'4A). FIRST CHECK IF
* ERROR > WINDOW
*
*****
*
* SUB THP1
* SACL THP3 : ERROR - WINDOW -> THP3
* BEZ AGC2

```

```

*****
* ERROR < WINDOW => CHECK IF ERROR > -WINDOW, IN WHICH CASE, ZERO THE
* ERROR. FIRST, ZERO THE ERROR (I.E. ASSUME ERROR > -WINDOW) AND MODIFY IF
* WRONG ASSUMPTION.
*
*****
*
* LACK ARL 0
* SAR ARL,TMP3 ; ASSUME ERROR > -WINDOW
*
*****
*
* CHECK ASSUMPTION
*
*****
*
* ARL TMP1,1 ; ERROR + WINDOW -> ACC
* BEZ AGC2 ; ASSUMPTION IS RIGHT
*
*****
*
* ERROR < -WINDOW => TMP0=(ERROR+WINDOW) -> TMP3
*
*****
*
* SACL TMP3
* LT TMP3
* MPY TMP0
* PAC
* SACL TMP3
*
*****
*
* AT THIS POINT, THE WEIGHTED WINDOWED ERROR IS CONTAINED IN TMP3. WE
* CONSIDER IT AN S.15 NUMBER AND USE IT TO UPDATE THE ACC GAIN ALPHA.
* FIRST, WE DETERMINE WHETHER TO SLEW OR NOT. IF THE ERROR IS LARGER THAN
* 1E40H OR SMALLER THAN FFE7H, GO INTO SLEWING MODE BY SETTING ERROR TO
* 7FFH or 800H RESPECTIVELY. OTHERWISE, LEAVE IT UNCHANGED.
*
* ALSO SET STATZ(7) APPROPRIATELY TO FREEZE THE UPDATE OF THE EQUALIZER
*
* STATZ(7)=1 UPDATE EQUALIZER
*
* STATZ(7)=0 FREEZE EQUALIZER
*
*****
*
* AGC2 LAC ONE,7
* OR STATZ ; ASSUME EQUALIZER UPDATE
* SACL STATZ
*
* AGC2 LAC TMP3
* LUPK 1

```

```

*
* SUB POSSN ; DO NOT SLEW
* AGC3
*
* LUPK 0 ; ENTER SLEW MODE
* LAC ONE,15
* SUB ONE
* SACL TMP3 ; TMP3 <- 7FFF
*
*
* LACK 7FH
* AND STATZ ; FREEZE EQUALIZER UPDATE
* SACL B
* AGC4
*
*
* AGC3 AND POSSN ; ACC <- TMP3
* AND NEGSH
* LUPK 0
* BEZ AGC4 ; DO NOT SLEW
*
* LAC ONE,15 ; ENTER SLEW MODE
* AND ONE
* SACL TMP3 ; TMP3 <- 8000
*
*
* LACK 7FH
* AND STATZ ; FREEZE EQUALIZER
* SACL STATZ
*
*****
*
* THE FOLLOWING LINES UPDATE THE GAIN ALPHA USING AN EXPONENTIAL INTEGRATOR
*
* ALPHA = ALPHA*(1-K*ERROR) (ERROR = TMP3)
*
* WHERE ALPHA IS OF FORMAT S7.8, ERROR IS S0.15, AND K = 0.5.
*
* ALPHA * ERROR: S7.8 * S.15 = S7.24.
*
* BY KEEPING ACC WITHOUT LEFT SHIFT THE MULTIPLICATION BY K IS
* ACCOMPLISHED.
*
* ALPHA IS UPPEXBOUNDED TO 35.73 IN S7.8
*
*****
*
* AGC4 LACK MAXVALP
* TBLR TMP0
* ZALH ALPHA ; ERROR -> T
* LT TMP3
* MPY ALPHA ; ALPHA (1 - 0.5*ERROR) -> ACC
* SACL ALPHA
*
*****
*
* CHECK IF ALPHA > MAX ALPHA

```



```

*****
*
* DECREMENT HYSTERESIS COUNTER
*
*****
*
* EDI2 BV EDI21 ; CLEAR OVERFLOW BIT
* EDI21 ZALH HYST ; HYSTERESIS COUNTER
* SUBH THPS ; THPS = 1927 = 32768/15
* SACH HYST
*
*****
*
* IN CASE OF OVERFLOW, DECLARE LOSS OF ENERGY DETECT
*
*****
*
* BV EDI02
* RET
* LACK ORFH
* AND STARD
* SACL STARD
* RET
*
*****
*
* FOLLOWING LINES ARE EXECUTED IF AFE GAIN IS HIGH, BUT NO ENERGY DETECT.
* CHECK IF ALPHA < 21.28 (I.E., RECEIVE LEVEL > -43.5 DBM) AND INCIDENT
* HYSTERESIS COUNTER IF IT IS, OTHERWISE, EXIT.
*
*****
*
* EDI01 LACK THRES2 ; 21.28 IN S7.8
* TBLR THPO
* LAC THPO
* SUB ALPHA
* BLZ EDI3
*
*****
*
* ALPHA < 21.28 => INCREMENT HYSTERESIS COUNTER
*
*****
*
* BV EDI011 ; CLEAR OVERFLOW BIT
* EDI011 ZALH HYST
* ADDH THPS ; THPS CONTAINS INC. X-OF
* SACH HYST
*
*****

```

```

*****
*
* DETECT BIT STARD(L3) = 1.
*
*****
*
* BV EDI04 ; IN CASE OF OVERFLOW, SET ENERGY
* RET
* EDI04 LAC ONE,6
* OR STARD
* SACL STARD
* RET
*
*****
*
* IF AFE GAIN STAGE IS BYPASSED, CHECK LEVEL OF ALPHA
*
*****
*
* EDI1 LACK THRES3
* TBLR THPO
* LAC THPO
* SUB ALPHA
* BLZ EDI3
*
*****
*
* IF ALPHA > THRES3 (20.09 IN S7.8) THEN TURN AFE GAIN STATUS WORD BIT ON.
*
*****
*
* LACK ORFH
* OR STARD
* SACL STARD
* LACK OL4h
* SACL ON
* RET
* EDI3
*
*****
*
* ROUTINE FOR SWITCHING THE AFE ON/OFF
*
*****

```

```

*****
*
* ZERO INAD ENERGY REGISTER
*
*****
*
* SWITCH ZAC AWESR
*
* SACL
*
* LACK O10A ; MASK OFF UNWANTED BITS
* AND ON
* BZ AFE0FF
*
*****
*
* CHECK IF THE GAIN SHOULD BE ON
*
*****
*
* LACK OFH ; MASK OFF THE AFE ON BIT
* AND ON
* SUB ONE ; DECREMENT THE COUNTER
* BZ SWITCH
*
* SACL ON ; SAVE ON VALUE
* LACK O10A ; LOAD THE AFE ON BIT
* OR ON
* SACL ON ; RESTORE AFE ON BIT
* RET
*
* SWITCH SACL ON ; RESET THE ON VALUE TO ZERO
* LACK THRESA ; LOAD THE NEW ALPHA VALUE
* TLR ALPHA ; RESET ALPHA TO 5.05
* RET
*
* AFE0FF LAC ON ; DECREMENT THE COUNTER
* SUB ONE
* BZ SWITCH2
*
* SACL ON
* RET
*
* SWITCH SACL ON ; LOAD NEW ALPHA VALUE
* LACK THRESA ; RESET ALPHA TO 8.98 IN 57.8
* TLR ALPHA
* RET
*

```