Digital Signal Processing Based
Biotelemetry Receivers

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SENSORS 2000!
NASA Ames Research Center

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Abstract

This is an attempt to develop a biotelemetry receiver using digital signal processing technology and techniques. The receiver developed in this work is based on recovering signals that have been encoded using either Pulse Position Modulation (PPM) or Pulse Code Modulation (PCM) technique. A prototype has been developed using state-of-the-art digital signal processing technology. A Printed Circuit Board (PCB) is being developed based on the technique and technology described here. This board is intended to be used in the UCSF Fetal Monitoring system developed at NASA. The board is capable of handling a variety of PPM and PCM signals encoding signals such as ECG, temperature, and pressure. A signal processing program has also been developed to analyze the received ECG signal to determine heart rate. This system provides a base for using digital signal processing in biotelemetry receivers and other similar applications.
Acknowledgments

We are grateful to NASA Ames Research Center for providing an opportunity to carry out the investigations outlined in this report. This work is a direct result of NASA Ames, San Jose State University Consortium NCC2-5173. We acknowledge assistance in implementations from Faranak Nekoogar of San Jose State University.

Avtar Singh, San Jose State University
John Hines and Chris Somps, NASA ARC
1. Introduction

Biotelemetry is a process by which physiological information or signals are transferred from one remote location to another, typically using radio frequency links. In today’s medical world, it is important to be able to transfer the biomedical signals to a remote location for non-interfering investigations and further processing. The importance of biotelemetry becomes obvious when we consider monitoring life in remote or inaccessible locations such as an astronaut in space or a baby in mother’s womb. The biomedical signals at the source are encoded, modulated and then transmitted. At the receiver end, the signals are decoded, displayed, and analyzed to extract diagnostic information for evaluation.

1.1 A Biotelemetry System

A biotelemetry system is shown in Fig. 1.1. There are two parts of this system, the transmitter and the receiver. The transmitter picks up a biomedical signals using sensors. These signals are processed using analog electronics to amplify and/or filter. The
processed signals are encoded and modulated to generate a signal suitable for wireless transmission. The transmitted signal, as received by the receiver, is demodulated, decoded and processed to recover the encoded signals. The recovered signals may be processed to generate diagnostically important information such as heart rate from an ECG signal.

1.2 A Multichannel, Multiple Subject Biotelemetry System

In a multichannel system, signals more than one are handled by the system. In a multiple subject system, signals from more than one subject are transmitted and received. By using an appropriate encoding scheme, a single pair of transmitter and receiver may be used to
handle a multichannel system. In order to realize a multiple subject biotelemetry system, the single subject multichannel systems can be duplicated as shown in Fig. 1.2. Such a configuration requires a unique transmission frequency, yet the same type of electronics, for each subject. This is generally how a multiple subject system is realized when analog signal processing techniques are used for the implementation.

1.3 Requirements of an Advanced Biotelemetry System

An advanced biotelemetry system should provide a multichannel and multiple subject capability. In addition to this basic requirement, it should provide programmability so that a given system can be used for different applications. The small size and minimum power are crucial for a portable system. Another requirement for a portable system is that it provide signal processing capability. When these requirements are collectively evaluated for implementation, a system based on a digital signal processor (DSP) is the one that has the best potential.

1.4 Biotelemetry Implementation Techniques

Current Biotelemetry employs radio frequencies to transmit and receive signals [1]. The signals are encoded using time or frequency division multiplexing [2]. The various techniques that have been developed, are available in literature and many have been implemented in commercial systems. For the purpose of saving power and space, these implementations are almost always based on analog circuits. The availability of low power microcontrollers has created interest among the researchers to use these devices to design biotelemetry systems. One such effort at NASA is by Jeutter [3] to use a low power microcontroller to design a biotelemetry transmitter. The inclusion of a processor in the transmitter and the receiver opens up enormous possibilities to use biotelemetry to provide flexible and controllable operation. A digital signal processor provides programmability to
design a flexible system as well as the signal processing capability. It still may not match the space and power requirements of a microcontroller device to design a transmitter, but it certainly is a good candidate to design a receiver where these requirements are not as stringent.

1.5 Goal of the Project

The goal of the research undertaken here is to develop a digital signal processor based architecture that can be used to receive multichannel biomedical signals from a class of biotelemetry transmitters/demodulators based on Pulse Position Modulation (PPM) as well as Pulse Code Modulation (PCM) encoding techniques. The PPM and PCM signals are to be decoded to generate the encoded biomedical signals. In the case of an ECG signal, it is to be further processed to extract heart rate. The system should allow handling of biotelemetry receivers based on the two pulse based encoding schemes by simply changing the software.
2. Design of a DSP Based Biotelemetry Receiver

In this chapter, we will develop a digital signal processor based architecture that can be used to implement a multichannel biotelemetry receiver. The Pulse Position Modulation (PPM) and Pulse Code Modulation (PCM) techniques used in such systems will be discussed along with the approach that will be used to decode these to recover the biomedical signals. An implementation approach for the biotelemetry receiver will also be described.

2.1 A DSP Based Biotelemetry Receiver

The block diagram shown in Fig. 2.1 can be used to implement a DSP based biotelemetry receiver. The DSP device receives the demodulated signal as obtained from the demodulator and analog processing circuits. The DSP device can be programmed to
decode the received signal by inverting the process of encoding used in the transmitter and thus generate the corresponding biomedical signals. The decoded signals are presented to a D/A converter to generate analog signals. The input to the DSP is one of the PPM or PCM signals discussed in the next section.

2.2 Modulation Techniques for Multichannel Biotelemetry Systems

The receiver developed in this project can be used with a variety of pulse modulation based biotelemetry systems. The two types of modulation schemes PPM and PCM are considered in this section. Both these schemes can be used to encode one or more biomedical signals.

2.2.1 Pulse Position Modulation (PPM)

A PPM signal that can be used to encode a single channel is shown in Fig. 2.2. The position of a pulse relative to the previous pulse (interval t1) encodes the sample value of the input signal. The nominal or the average value of the interval t1 may be used to define the average sampling rate. For instance in Fig. 2.2, this rate is 5 KHz.

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Function</th>
<th>Duration (uSec)</th>
</tr>
</thead>
<tbody>
<tr>
<td>t1</td>
<td>Encodes Signal 1</td>
<td>1900</td>
</tr>
<tr>
<td>Sync Interval</td>
<td></td>
<td>100</td>
</tr>
<tr>
<td>Total</td>
<td>Sampling Interval</td>
<td>2000</td>
</tr>
</tbody>
</table>

Figure 2.2. A PPM Signal for Encoding a Biomedical Signal
A PPM signal that encodes two signals along with providing a fixed sampling rate is shown in Fig. 2.3. Such a signal requires a sync signal (two pulses) to mark the beginning of a cycle for encoding two or more signals. As shown in Fig. 2.3, t1 encodes one signal, and t2 encodes the other. The time interval t3 is simply needed to keep the sampling interval t1+t2+t3 constant to provide a fixed sampling rate. In the example shown in Fig. 2.3, the fixed sampling rate is 5 KHz.

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Function</th>
<th>Duration (μSec)</th>
</tr>
</thead>
<tbody>
<tr>
<td>t1</td>
<td>Encodes Signal 1</td>
<td>1000</td>
</tr>
<tr>
<td>t2</td>
<td>Encodes Signal 2</td>
<td>800</td>
</tr>
<tr>
<td>t3</td>
<td>Compensation Interval</td>
<td>1700</td>
</tr>
<tr>
<td>Each Pulse Interval</td>
<td></td>
<td>100</td>
</tr>
<tr>
<td>Sync Interval</td>
<td>Sampling Interval</td>
<td>3 x 100</td>
</tr>
<tr>
<td>Total</td>
<td></td>
<td>4000</td>
</tr>
</tbody>
</table>

In a modified PPM encoding technique the interval t3 as shown in Fig. 2.3 is eliminated, generating a PPM signal shown in Fig. 2.4. This means that the sampling interval and hence the sampling rate is going to vary depending on the amplitude of the signal being encoded. The advantage of such encoding system is that it can be used to save power in the transmitter. This type of encoding is used in the Fetal Monitoring System developed at NASA in conjunction with UCSF’s Fetal Treatment Center. This system has been further modified to encode three signals by superimposing a third very low frequency signal on
one of the intervals t1 or t2. The system is typically used to encode ECG, temperature, and pressure signals, temperature being the lowest frequency signal is combined with the highest frequency ECG signal for encoding interval t2.

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Function</th>
<th>Duration (uSec)</th>
</tr>
</thead>
<tbody>
<tr>
<td>t1</td>
<td>Encodes Signal 1</td>
<td>800</td>
</tr>
<tr>
<td>t2</td>
<td>Encodes Signal 2. Signal 3</td>
<td>800</td>
</tr>
<tr>
<td>Each Pulse Interval</td>
<td></td>
<td>80</td>
</tr>
<tr>
<td>Sync Interval</td>
<td></td>
<td>3 x 80</td>
</tr>
<tr>
<td>Total</td>
<td>Sampling Interval</td>
<td>~2000</td>
</tr>
</tbody>
</table>

Figure 2.4. A PPM Signal used in UCSF Fetal Monitoring Biotelemetry System

2.2.2. Pulse Code Modulation (PCM)

The PCM signal is a set of binary digits representing the code word for the quantized value of the sample. PCM offers improved performance over other modulation techniques in low-signal-to noise environments, since the receiver decoder has to detect only the presence of a pulse [1]. Figure 7.2 shows the PCM signal, that is used as the input to the decoder circuit.
As shown in Fig. 2.5 the received PCM signal is a set of pulses consisting of two parts. The first part is the preamble which is a series of 12 "ones" followed by space with zero indicating that the encoded data is to follow. The second part contains the seven bits of address that starts with a "one" followed by the seven address and eight bits of data. Fig. 2.6 shows the format of the PCM signal. The address bits are there for signal source or type identification. The data bits encode the signal value.

![Figure 2.5. Pulse Code Modulation Signal](image)

<table>
<thead>
<tr>
<th>Preamble</th>
<th>Space Zero</th>
<th>1</th>
<th>X</th>
<th>X</th>
<th>X</th>
<th>X</th>
<th>X</th>
<th>D</th>
<th>D</th>
<th>D</th>
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<tbody>
<tr>
<td>A0</td>
<td>A6 D0</td>
<td></td>
<td></td>
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</table>

![Figure 2.6. Format of the Encoded PCM Signal](image)

To encode a signal the Manchester Coding scheme is used. In this method, each bit is represented by two successive pulses of opposite polarity. As shown in Fig. 2.7, a bit value 1 is represented by 10 and a bit value 0 by 01 pulse sequence.

![Figure 2.7. Manchester Coding Scheme](image)
The encoded PCM signal is further processed to obtain a train of narrow pulses representing transitions in the PCM signal. This signal called Edge Based PCM signal is the one that is input to the system. It is shown in Fig. 7.5.

![PCM and Edge Based PCM Signal](image)

**Figure 2.8. PCM and Edge Based PCM Signal**

The signal shown in Fig. 2.8 represents "11001001" binary code. It may be noted that the encoded signal is characterized by edges separated by an interval $\tau$ or $2\tau$. Therefore by measuring the times between same polarity adjacent edges a decoding scheme can be designed to extract the encoded signal.

2.3 A DSP Based Decoding Scheme for the PPM Receivers

The schematic diagram in Fig. 2.9 shows how a DSP device can be used to decode a PPM signal to recover encoded biomedical signals. The decoding requires measurements of time intervals in a PPM signal. The DSP device timer can be used for time measurement. To initiate the measurement process, the pulses in the PPM signal can be used to generate interrupt signals for the DSP device, which then are used to start or terminate the timer. This approach avoids using an A/D converter to handle the PPM signal, but it requires
that the DSP device be fast enough so as not to miss a pulse or introduce time measurement error. Typically in a PPM based system, the pulse duration is 100 μS. Thus if the DSP device can respond in 100 nS (which is the case for TMS320C50 digital signal processor), the time measurement error and hence the signal decoding error due to DSP device’s interrupt response will be less than 0.1%. The signal decoding accuracy also depends upon the timer used to measure the time intervals. In the TMS320C50 digital signal processor, the timer runs at 10 MHz, thus providing a capability to make a time measurement with an error limited to 100 nS or another 0.1% decoding error. Therefore, a decoding scheme based on the TMS320C50 signal processor can be designed so that the decoding error is limited to 0.2%.

2.4 A DSP Based Decoding Scheme for the PCM Receivers

The edge-based PCM signal encodes the signal type (or source) and its value using Manchester coding scheme. To decode this signal, we need to invert the encoding process which requires measuring time intervals between the pulses. To accomplish this objective, similar to PPM receivers, the signal can be applied to an interrupt input of the DSP.
The DSP Interrupt is activated each time a rising edge occurs in the applied signal as shown in Fig. 2.10. Based on the intervals between the interrupts which are determined using the DSP timer, the signal can be decoded for its type and value. As shown in Fig. 2.11, the measured time intervals between the interrupts are first converted to a train of alternate ones and zeros starting with a one. An interval \( \tau \) represents a single 1 or a 0, whereas an interval \( 2\tau \) represents two 0s or two 1s. This encoded signal is further decoded to the signal value by replacing a 10 sequence with a 1 and a 01 sequence with a 0 starting.
from the data start bit which is a 1. This process yields the data representing 7 address bits and 8 data bits. The data value can be further processed for the desired parameters. For instance if the received signal is an ECG, it can be processed to determine the associated heart rate.

To detect the preamble and the period of zeros, the interval between interrupts is calculated and compared to interval “τ” (Fig. 2.5). 24 “τ” intervals represent 12 pulses of preamble and if the interrupt does not happen for a relatively long time (approximately 8τ) the zero period is detected. The flowchart of Fig. 2.12 shows the details of implementing the PCM decoding technique discussed here.
Figure 2.12 Software Flowchart using Manchester Coding
In this chapter, we discuss the implementation of a DSP based biotelemetry receiver whose architecture was considered in the previous chapter. Major hardware sections of the receiver, such as the digital signal processor and the D/A converter for generating analog signals are also described.

### 3.1 A DSP Based Biotelemetry Receiver Implementation

The block diagram in Fig. 3.1 shows the system used for implementation. The PPM or the PCM signal is first processed using an isolation circuit before it is applied to the interrupt system of the signal processor. The DSP device is interfaced to a four channel digital to analog converter so that signals can be generated for analog display monitoring devices. The signal processor in the system is the TMS320C50 device [4]. An EPROM device provides storage for the board operating system as well as the decoding software. A serial port provides access to the signal processor from a PC for debugging purpose. The EPROM provides a very basic debugging software as part of the board’s operating system. The complete schematic for the system is available in Appendix I.

### 3.2 TMS320C50 Digital Signal Processor

The heart of the receiver is the digital signal processing device TMS320C50 [4]. This processor provides signal processing instructions for fixed-point calculations. It is a static device that can run with any clock with period as low as 100 nS. A low power version of the device that operates from 3.3V, uses only 35 mA of current at full speed operation. This DSP device also provides two power saving modes, sleep mode (5 uA) and
peripheral mode (23 mA). Most instructions execute in one clock period. The device provides 9K words of on-chip memory for programs and data. To implement the decoding of the PPM and PCM signals, an on-chip 20 bit timer is available for time measurements.

3.3 The Digital to Analog Converter

In order for the DSP to generate the recovered biomedical signals, a four channel parallel digital to analog converter is used. It is an 8 bit converter that is interfaced directly to the signal processor without using any additional hardware. This limits the I/O ports in the system only to the ones on the D/A converter. The four channels have I/O addresses 0, 1, 2, and 3.
3.4 Software for the Receiver

Two types of software programs are stored in the EPROM. One is the software for decoding PPM and PCM signals to generate the encoded biomedical signals. The other software allows to provide debugging capability using a PC connected to the RS232 connection provided on the board. This software provides standard debugging functions similar to those available on the development board from Texas Instruments [5]. The EPROM is socketed so that future revisions of the board software are easy to implement.

The decoding software listing for decoding the PCM signal is included in the Appendix B. The PPM decoding software is available in an earlier report [7].

3.5 A PCB for the Receiver

For our prototype we used a board from Texas Instrument that incorporates the TMS320C50 signal processor [5]. The board has other features suitable for general purpose software development. We utilized an on-board D/A converter to provide analog signal output for the computed analog signals. The board includes a complete PC based development software consisting of an Assembler, a Debugger, and a C Compiler. For debugging during development, the board can be accessed using a serial port on the PC. This is the development mode that was used in this project. Based on the prototype design a PCB is under development at NASA. This board is intended to be used in the UCSF Fetal Monitoring system [6]. However, the board is suitable for many applications requiring DSP calculations and producing analog signals.
4. Digital Signal Processing for the Received Biomedical Signals

In chapter 2, we described the signal processing needed to recover the biomedical signals by decoding the PPM or the PCM signal using the interrupt and the timer capability of the signal processor. Now, we consider digital signal processing techniques for processing the received ECG signal to determine the associated heart rate.

4.1 ECG Signal Processing for Heart Rate Determination

The most important information contained in an ECG signal is the associated heart rate. Determining the heart rate involves determining the time interval between QRS complexes. Therefore, we need a reliable algorithm to detect the QRS complexes so that the QRS interval can be determined to compute the heart rate. Fig. 4.1 shows the steps of operation that can be performed to determine the heart rate.

**Figure 4.1. ECG Signal Processing for Heart Rate**
A nonlinear transformation is used to enhance the QRS complex so that it can be detected reliably with a threshold detector. The transformation in our implementation uses absolute values of the first and second derivatives of the signal as follows

\[ y_1(n) = |x(n) - x(n-1)| \]  \hspace{1cm} (4.1)
\[ y_2(n) = |x(n-2) - 2x(n-1) + x(n)| \]  \hspace{1cm} (4.2)
\[ y_3(n) = y_1(n) + y_2(n) \]  \hspace{1cm} (4.3)

where \( x(n) \) refers to the ECG signal sample, \( y_1(n) \) is the absolute value of the first derivative, \( y_2(n) \) is the absolute value of the second derivative, and \( y_3(n) \) is the combined absolute first and second derivatives.

The transformed signal is filtered to remove high frequency noise components. To accomplish this we used a simple IIR filter as follows

\[ y_4(n) = \alpha(y_3(n) - y_4(n-1)) + y_4(n-1) \]  \hspace{1cm} (4.4)

where \( \alpha \), a number less than 1, is the IIR filter coefficient. Its value is chosen based on the smoothing effect that should be used to discard high frequencies. \( y_4(n) \) in the difference equation (4.4) denotes the filtered transformed signal.

A QRS complex is detected using a threshold detector. The threshold for the detector is determined by processing typical ECG signals by the above algorithm and determining the mean of half of the peak amplitudes of the filtered signals. This estimated threshold value is then used to detect the QRS complexes in a given ECG waveform.

The time interval between two complexes is the QRS interval. Finally, the heart rate (HR) in beats per minute (BPM) is computed using the formula

\[ HR = \frac{(\text{Sampling Rate} \times 60)}{\text{QRS Interval}} \]  \hspace{1cm} (4.5)
The Sampling Rate is determined from the time duration of a PPM or a PCM cycle or depends upon the modulation technique. If the PPM cycle duration keeps changing (as is the case for the Fetal Telemetry System), the sampling interval will keep changing. To produce a heart rate value accurate on an average, the computed heart rate can be filtered using a filter similar to the one in equation 4.4.

4.2 Signal Processing for Temperature Encoded with ECG

In the fetal monitoring biotelemetry system, ECG and temperature are encoded together in one of the time intervals of the PPM signal. This is done to save power in the transmitter. In the receiver, the two signals must be separated using appropriate filtering methods. The temperature relative to ECG is a low frequency signal. Therefore, lowpass and highpass filters can be used to separate the two signals. The problem with this approach is that an accurate determination and implementation of the cut-off frequencies of these filters is a difficult task.

The DSP based implementation of the receiver allows using a scheme that avoids using spectral filters to separate the two signals. The approach we used is to use the signal between the QRS complexes to generate the temperature signal. This approach requires the same QRS detection scheme as for the heart rate calculations and therefore does not require a new implementation. The temperature signal generated, is filtered with a lowpass filter to reject any noise due to errors or delays in the QRS detection scheme. This filter can be a simple IIR or FIR filter and does not need to satisfy strict cut-off frequency requirement. The temperature signal then can be subtracted from the combined signal to generate an ECG only signal.
5. Conclusion and Recommendations

This work has shown that a DSP device can be successfully used to implement decoding operations in a biotelemetry receiver system based on a Pulse Position Modulation (PPM) or a Pulse Code Modulation (PCM) technique. A prototype system based on a commercially available DSP board has been developed. This system receives a PPM or a PCM signal and generates the corresponding biomedical signals. The signals are available at the outputs of a four channel D/A converter. An ECG signal is further analyzed to determine the heart rate which then is output as an analog signal on one of the D/A channels. It is important to note that the same hardware has been used to decode a variety of PPM as well as PCM signals by simply invoking the appropriate software.

A printed circuit board (PCB) is under development at NASA. This board will be usable in the existing UCSF Fetal Biotelemetry system as well as in a PCM system under development.
References


NOTE
- There are 5 separate grounds in the box: VEE, VDD, VEE-OUT, VDD-OUT.
- Component 979 is 120VAC. Use 25 volts for VEE and 100 volts for VDD.
- All other components use 5 volts as VDD at the power plane.
- A 10uF capacitor is needed between the VEE and VDD of any component.

The following describes the power plane and GND plane for different components:
- VDD
  - 12V (5 volts)
  - 6V (5 volts)

Comparator 01A is used for testing purposes.
Appendix B

DSP Based PCM Receiver Decoding Software
PULSE CODE MODULATION BIOTELEMETRY DECODER (PCMBT)

Source File: pcm_test.asm
Description: This program accepts modulated PCM signal at INT1 and decodes the signal to generate ECG, Temperature, HR indications at the output ports of an external D/A converter.

History: Last update done on 4/8/97

Defined variables

.mmregs ;Memory-mapped registers of CSX

Interrupt vectors

.ps 0802h
1: B ISR1 ;02; External INT1
.ps 0808h
.T: B TIMER ;08; Timer Interrupt
.ps 080ch

Data memory allocations

.ds 01000h
.nal
  .word 0
  .word 24 ;Counter for the Sync. pulses
  .word 1
  .word 1
.h
  .word 0 ;Memory high
.l
  .word 0 ;Memory low
.bt
  .word 4096 ;Memory locations for shifting purposes to decode data
  .word 0
  .word 0
  .word 0
  .word 0
  .word 0
  .word 0
  .word 0
  .word 0
  .word 0
  .word 0
  .word 0
  .word 0
  .word 0
  .word 0
  .word 0
  .word 0
.word 0 ;Memory location to store the decoded data
f .word 255 ;00FFh
buf .word 0 ;Memory locations for shifting purposes to decode the address
buf .word 0
buf .word 0
buf .word 0
buf .word 0
buf .word 0
buf .word 0
buf .word 0
buf .word 0
buf .word 0
buf .word 0
buf .word 0
buf .word 0
buf .word 0
buf .word 0
buf .word 0
g: .word 0 ;0=preamble (sync pulses+zero period)
1 .word 0 ;1=data
g1: .word 0 ;0=12 sync pulses
2 .word 0 ;1=zero period
vCnt: .word 0 ;Interval count
vCnt1: .word 1110 ;1.5t
0Intv: .word 0 ;Interval for zero period
aIntv: .word 0 ;Interval for data
tch: .word 0 ;glitch

Computation variables
m: .word 0
ml: .word 0
ESH: .word 50 ;Threshold for QRS detection
I:  .word 0
IM1: .word 0
tr:  .word 0
nt:  .word 0
1  .word 0
per: .word 0
perml: .word 0
ha:  .word 4000h ;Alpha = .5 for the filter
pl:  .word 0

gnal buffers

lbuf: .word 0,0,0,0,0,0,0,0
      .word 0,0,0,0,0,0,0,0
      .word 0,0,0,0,0,0,0,0
      .word 0,0,0,0,0,0,0,0
      .word 0

h0 .set 0000h
h1 .set 0001h
h2 .set 0002h
h3 .set 0003h

                    .ps 0a00h ;Program start address
                    .entry

 *************************************************

ain Program

*************************************************

RT:
    ldp    #0 ;Select page 0
    setc   INTM ;Disable interrupts
    call   initC5X ;Init C5X
    splk   #0bh, IMR ;Unmask INT1,INT2 and TINT
    ldp    #0 ;Select page 0
    call   initTimer ;Init the timer
    splk   #1fh, IFR ;Reset pending interrupts
    clrc   INTM ;Enable interrupts

    t: nop ;Wait for interrupts to occur

    b    wait

---------------------------------------------------
inIC5X

---------------------------------------------------
tC5X:
    ldp    #0 ;Select page 0 for direct addressing
nitTimer

The timer is initialized to #FFFFh which represents 50 msec, and each count is equal to 0.8 Usec.

tTimer:

<table>
<thead>
<tr>
<th>ldp</th>
<th>#0</th>
<th>Select page 0 for direct addressing</th>
</tr>
</thead>
<tbody>
<tr>
<td>splk</td>
<td>#0ffffh,PRD</td>
<td>Init Timer with 0ffffh (50 msec)</td>
</tr>
<tr>
<td>splk</td>
<td>#2fh,TCR</td>
<td>and start it</td>
</tr>
<tr>
<td>ret</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

Interrupt Service Routine ISR1

This routine reads the 16 MSBs of the timer, computes the count by which the timer has counted down from #ffffh and saves the computed interval in the signal buffer to decode the PCM signals. A flag is used to distinguish the preamble pulses from address and data pulses. Before returning back the timer is reset to ffffh.

1:

<table>
<thead>
<tr>
<th>ldp</th>
<th>#Signal</th>
<th>Location for data page</th>
</tr>
</thead>
<tbody>
<tr>
<td>lacc</td>
<td>flag</td>
<td>flag=0 represents the preamble</td>
</tr>
<tr>
<td>sub</td>
<td>#1</td>
<td>flag=1 represents data</td>
</tr>
<tr>
<td>bcnd</td>
<td>IntvChk,EQ</td>
<td>If flag=1, check the intervals to decode data</td>
</tr>
<tr>
<td>lacc</td>
<td>flag1</td>
<td>flag1=0 represents the sync pulses</td>
</tr>
<tr>
<td>sub</td>
<td>#1</td>
<td>flag1=1 represents the zero interval</td>
</tr>
</tbody>
</table>
bcnd ZeroChk,EQ ;If flag1=1, check for zero period
ldp #0 ;Data page = 0
opl #10h, TCR ;Stop the Timer
lacc TIM ;Read the Timer
splk #2fh, TCR ;Reload & restart the Timer
neg ;Compute the timer-count-down

ldp #Signal
sacl IntvCnt ;Save it temporarily
sub IntvCnt1 ;Is this preamble interval?
bcnd skip2ex, GEQ ;If Yes, check for the zero period, if not,
lacl cnt ;Exit the service routine
sub #1h ;Read the timer
sacl cnt ;Decrement the counter
bcnd skip2ex, GT ;Exit the service routine if the count is
b ;not 0.

splk #1, flag1 ;If the count is 0, flag1=1.
b skip2ex ;Exit the service routine uncoditionally.

oChk:
ldp #0 ;Reading the timer & computing the timer
opl #10h, TCR ;count down.
lacc TIM
splk #2fh, TCR
neg

ldp #Signal
sacl ZeroIntv ;Using the value of the timer to check
sub ZeroCnt ;against the constant interval 3.7 msec.
bcnd skip2ex, GEQ ;If zero period is detected, flag1=data
splk #1, flag ;is ready
b skip2ex

vChk:
ldp #0 ;Reading the timer & computing the timer
opl #10h, TCR ;count down.
lacc TIM
splk #2fh, TCR
neg

ldp #Signal
sacl IntvCnt ;Checking the value of the timer against
sub IntvCnt1 ;1.5t.
bcnd t1Interval, LT ;Check if the interval is less than or
bcnd t2Interval, GT ;greater than 1.5t.
bcnd skip2ex, EQ

nterval:
ldp #Signal ;Decoding the data based on having
lacc memh ;intervals less than 1.5t.
rpt #15 ;Shift accumulator 16 times.
sfl or meml ;Store a 32 bit number in accumulator.
sfl
or BT ;Shift left, store "BT" in "meml"
sach memh
sacl meml
lacc BT
compl ;Complement "BT"
and BT1 ;Use LSB of "BT"
sacl BT
b FinalChk ;Check if 32 bits are in the accumulator.

interval:

ldp #Signal ;Decoding the data based on the intervals
lacc memh
rpt #15
sfl or meml ;Store a 32 bit number in accumulator
sfl or BT ;Shift left and store "BT"
sfl or sach memh
sacl meml
lacc memh ;Storing & shifting is repeated
rpt #15
sfl or meml
sfl or BT
sfl or sach memh
sacl meml
lacc BT ;Complement "BT"
compl and BT1 ;Use LSB of "BT"
sacl BT

alChk:

ldp #Signal ;Checking if the MSB in accumulator
lacc memh
sub lstbt ;is at least 1000h, so the LSB is
bcnd Shift,GEQ ;transferred to MSB.
splk cnt,24 ;Set the counter to 24.
splk flagl,1 ;Set the flagl to 1.
splk flag,1 ;Set the flag to 1.
b skip2ex ;Skip to exit.

ft:

ldp #Signal
lacc memh
rpt #15
sfl or meml ;Storing 32 bits in the accumulator.
sfr ;Starting from here acc. is shifted
sacl mem1 ;right each time and acch is stored
sach ml ;in m(i) and acch is stored in mem(i).
and   #0001h   ; (i) starts at 1 and ends at 16, so
sacl   mm1    ; this process will be repeated 16 times
lacc   m1     ; The bit of interest is stored in mm
rpt    #15    ; locations.
sfl    
or     mem1

rpt    #1    ; Second Bit.
sfr    sacl   mem2
sach   m2
and    #0001h
sacl   mm2,1
lacc   m2
rpt    #15
sfl    or     mem2

rpt    #1    ; Third Bit.
sfr    sacl   mem3
sach   m3
and    #0001h
sacl   mm3,2
lacc   m3
rpt    #15
sfl    or     mem3

rpt    #1    ; Fourth Bit
sfr    sacl   mem4
sach   m4
and    #0001h
sacl   mm4,3
lacc   m4
rpt    #15
sfl    or     mem4

rpt    #1    ; Fifth Bit.
sfr    sacl   mem5
sach   m5
and    #0001h
sacl   mm5,4
lacc   m5
rpt    #15
sfl    or     mem5

rpt    #1    ; Sixth Bit
sfr    sacl   mem6
sach   m6
and    #0001h
sacl   mm6,5
lacc   m6
rpt    #15
sfl    
or mem6
rpt #1 ;Seventh Bit
sfr sacl mem7
sach m7
and #0001h
sacl mm7,6
lacc m7
rpt #15
sfl or mem7
rpt #1 ;8th Bit.
sfr sacl mem8
sach m8
and #0001h
sacl mm8,7
lacc m8
rpt #15
sfl or mem8
rpt #1 ;9th Bit.
sfr sacl mem9
sach m9
and #0001h
sacl mm9
lacc m9
rpt #15
sfl or mem9
rpt #1 ;10th Bit.
sfr sacl mem10
sach m10
and #0001h
sacl mm10,1
lacc m10
rpt #15
sfl or mem10
rpt #1 ;11th Bit.
sfr sacl mem11
sach m11
and #0001h
sacl mm11,2
lacc m11
rpt #15
sfl or mem11
rpt #1 ;12th Bit.
sfr sacl mem12
sach   m12
and   #0001h
sacl  mm12,3
lacc  m12
rpt   #15
sfl
or   mem12

rpt   #1                           ;13th Bit.
sfr
sacl  mem13
sach  m13
and   #0001h
sacl  mm13,4
lacc  m13
rpt   #15
sfl
or   mem13

rpt   #1                           ;14th Bit.
sfr
sacl  mem14
sach  m14
and   #0001h
sacl  mm14,5
lacc  m14
rpt   #15
sfl
or   mem14

rpt   #1                           ;15th bit.
sfr
sacl  mem15
sach  m15
and   #0001h
sacl  mm15,6
lacc  m15
rpt   #15
sfl
or   mem15

rpt   #1                           ;16th Bit.
sfr
sacl  mem16
sach  m16
and   #0001h
sacl  mm16,7
lacc  m16
rpt   #15
sfl
or   mem16

lacc  mm1
or   mm2
or   mm3
or   mm4
or   mm5
or   mm6
or   mm7
or   mm8

;Starting from this part the logic "OR" operation is done on mm1 to mm16.
;Decoded Data.
sacl sigbuf

;Decoded Address.
sacl siglbuf

lacc mm9
or mm10
or mm11
or mm12
or mm13
or mm14
or mm15
or mm16

sacl sig1

call analogout ;Output the signal to DAC.

;Return with interrupts enabled
;and registers restored.

---

alog signal output routine (analogout)

;summing the signal samples are form ECG + Temp signal, this routine
;computes, QRS indication signal, Temperature, Heart Rate Indications
;and outputs these signals along with the original signal on 4 output
;channels of the D/A converter.

alogout:

ldp #Signal ;Select the direct addressing page

mar *,ar5 ;y0(n) = abs(x(n)-x(n-1))
lar ar5,#Siglbuf
lacc *+
sub *+
abs
sacl yOn
lacc *-
sub *-,-1
add *
abs
sacl y1n
add y0n ;y1(n) = abs(x(n-2)-2x(n-1)+x(n))
sacl y2n
sub y2nm1
sacl templ
lt Alpha
mpy templ
pac
add #8000h
addh y2nm1
sach y2n
lacc y2n
;if y2(n) > THRESH then QRSI = 7fffch
sub THRESH
bcnd SkipA, GT
else QRSI = 0

bcnd SkipB, LT
ret
sacl QRSI
b SkipB

A:
lacc #7ffch
sacl QRSI

B:
sub QRSIM1 ;If QRSI-QRSM1 =or< 0 then
bcnd SkipC,GT ;

lacc HRCtr
add #1
sacl HRCtr
b SkipD

C:
lacc HRCtr ;If QRSI-QRSM1 > 0 then HRInt = HRCtr
sacl HRInt ;

lacc #0
sacl HRCtr
lacc #29000 ;HR = (Sampling rate * 60)/HRInt.
rpt #15
subc HRInt

and #1ffh ;Limit HR to 511 BPM.

sacl HR ;Filter the HR.

D:
sub Hrm1
sacl templ

lt Alpha
mpy templ
pac
add #4000h
addh Hrm1
sach HR

mar *,ar5 ;Get temperature x
adrk #15
lacc * ;Filter the temperature
rpt #4 ;Temperature value reduced to eight bits
sfr
and #1ffh ;Limit Temp to 511.
sub Temperml
sacl templ
lt Alpha
mpy templ
pac
add #4000
addh Temperml
sach Temper

D:
dmov y2n ;Shift the signal arrays
dmov QRSI
dmov HR
dmov Temper

out sig1buf,DAch0 ;Output signal at channel 0 of D/A
out QRSI,DAch1 ;Output QRS indicator at channel 1 of D/A