

# ***Improving 32 Channel DTMF Decoders Using the TMS320C5x***

*Application  
Report*



# ***Improving 32 Channel DTMF Decoders Using the TMS320C5x***

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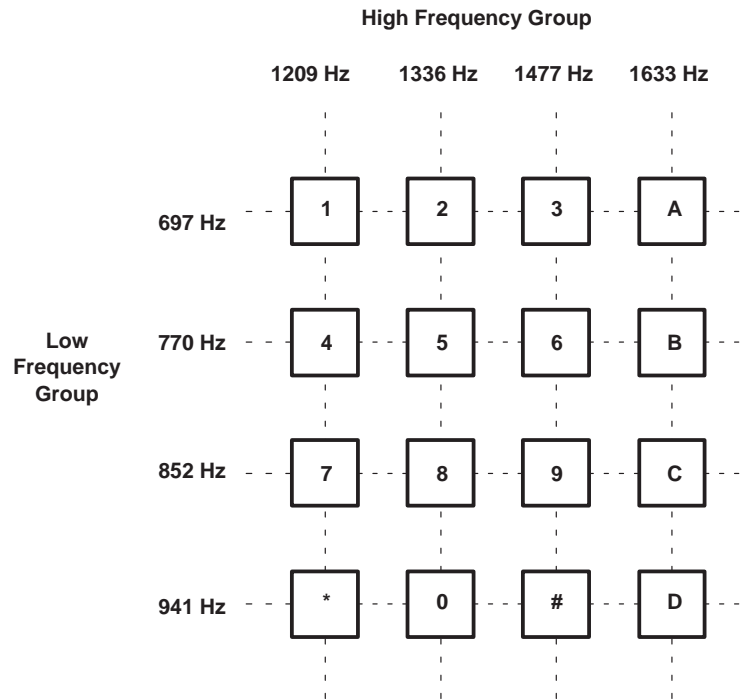
## ABSTRACT

PBX office switching systems use dual tone multifrequency (DTMF) signals since the signals serve as data entry terminals. DTMF uses a pair of frequencies to produce a tone. PBX systems use multiple DTMF chips to encode or decode the tones. PBX systems also process other functions such as voice compression or expansion and voice mail through their systems. Extending a PBX system to encompass these functions is expensive. DSPs increase the performance and flexibility of the PBX systems, while decreasing the cost. This application report discusses improvements to the multi-channel DTMF decoder using a TMS320C5x DSP<sup>1</sup>.

## DTMF TONE ENCODING

DTMF is a touch-tone signal comprised of two frequencies. One frequency is from the high frequency group, (1209 Hz, 1336 Hz, 1477 Hz, 1633 Hz) and one is from the low frequency group (697 Hz, 770 Hz, 852 Hz, 941 Hz). Figure 1 shows DTMF tone generation.

Pressing 1 on a 16-key pad generates a DTMF signal pair. This signal pair is 1209 Hz from the high frequency group and 697 Hz from the low frequency group.



**Figure 1. DTMF Touch-Tone Frequency**

The DTMF encoder uses the two frequencies to calculate a value. The DTMF encoder equation is:

$$Y(n) = 2\cos(\text{coef}) \times Y(n-1) + (-1) \times Y(n-2) + [\sin(\text{coef}) \times X(n-1)] \tag{1}$$

where  $\text{coef} = 360 \times \text{frequency/sample rate}$ .

The encoder equation uses the following initial data:

$$Y(0) = 0; Y(1) = \sin(x); X(1) = 1; \text{ and } X(n) = 0 \text{ for } X \neq 1.$$

The sample rate is 8 KHz.

## DTMF TONE DECODING

DTMF tone decoding uses codes that are more complicated than DTMF encoding. The decoding program uses a discrete Fourier transform (DFT) algorithm, known as Goertzel's algorithm. The algorithm is a series of second-order infinite impulse response (IIR) filters. The DTMF decoder equation is:

$$Y(n) = X(n) + 2\cos(\text{coef}) \times Y(n-1) + (-1)Y(n-2) \tag{2}$$

where  $\text{coef} = 360 \times \text{frequency}/\text{sample rate}$  and  $X(n)$  is the current sample value.  $Y(n-1)$  and  $Y(n-2)$  are feedback storage elements for the frequency point,  $k$ . The value of  $k$  is:

$$k = N \times \text{Freq}/(\text{Sample Rate}) \tag{3}$$

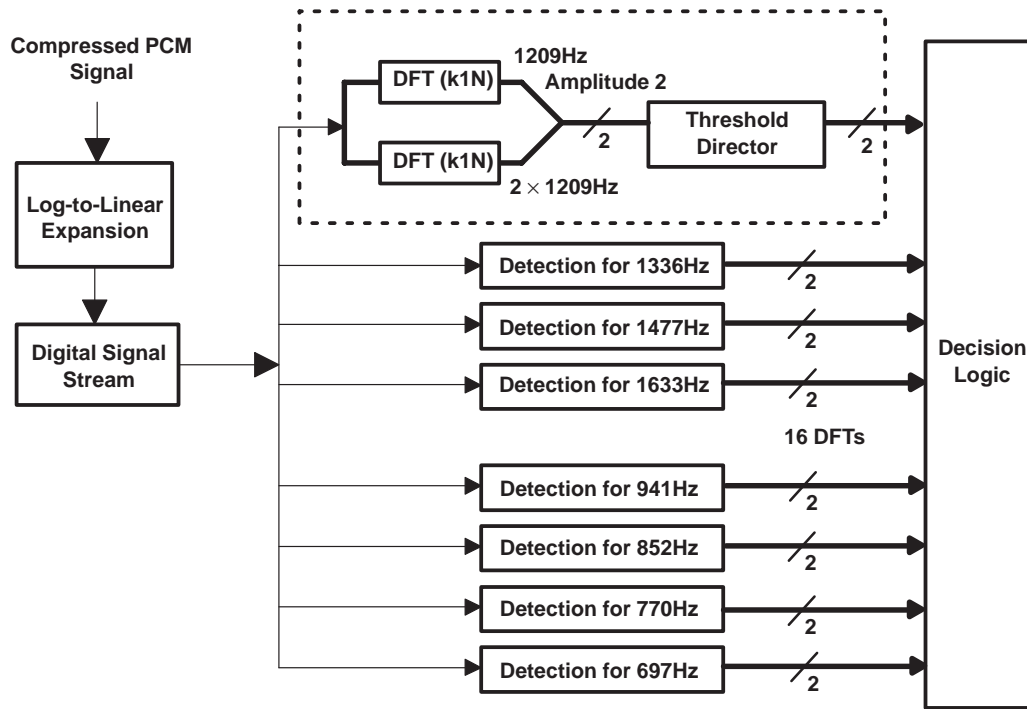
The decoder equation uses the initial data:

$$Y(-1) = 0, Y(-2) = 0, Y(-3) = 0, \dots \text{ for } n = 0, 1, 2, \dots, n-1.$$

The sample rate is 8 KHz.

Figure 2 shows a diagram of a 16-tone DTMF decoder. Input data decodes linearly and a code compresses several phone signals. A compressed pulse coded modulation† (PCM) signal goes through the log-to-linear expansion, then is put into a stream form. Once it is in a stream form, the samples go to the DFT algorithm.

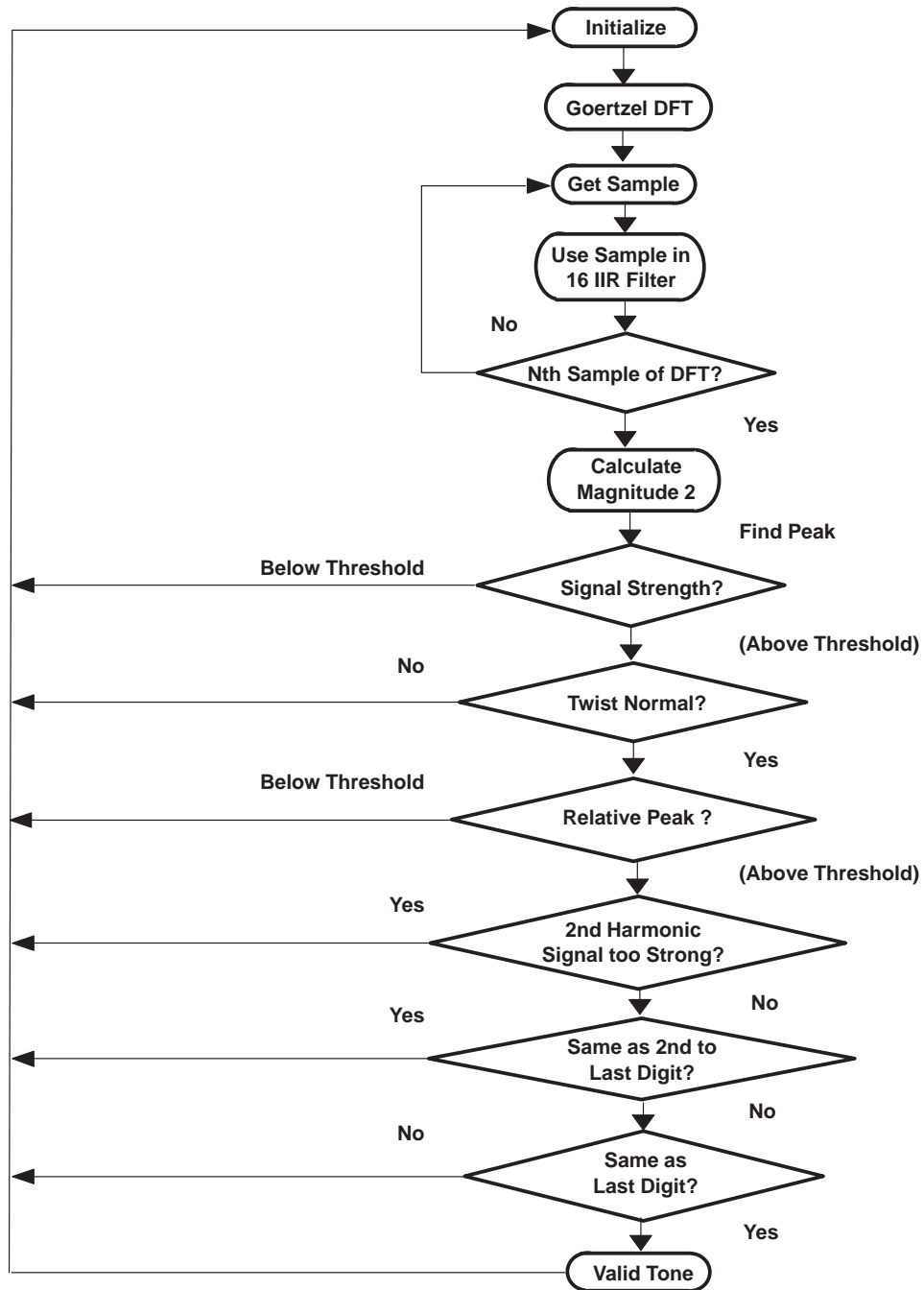
The samples are put into a parallel array containing the DFT algorithms. They are processed serially. There are sixteen algorithm for the tone detection of the DTMF frequencies and second harmonics. These second harmonics distinguish between speech and DTMF tones.



**Figure 2. Tone Decoder — 16 DFT Transforms**

† PCM is the most common form of digital waveform coding.





**Figure 3. DTMF Tone Detection**

Figure 3 shows the flow of data for DTMF tone detection. After processor initialization, the program sends the first sample to the 16 DFT IIR filters. The program compares the sample data with the thresholds to detect valid DTMF tones. A complete description of the tone decoder and DTMF tone detection is in the *TI DSP Application Book Volume 1*.

## PERFORMANCE CONSIDERATION

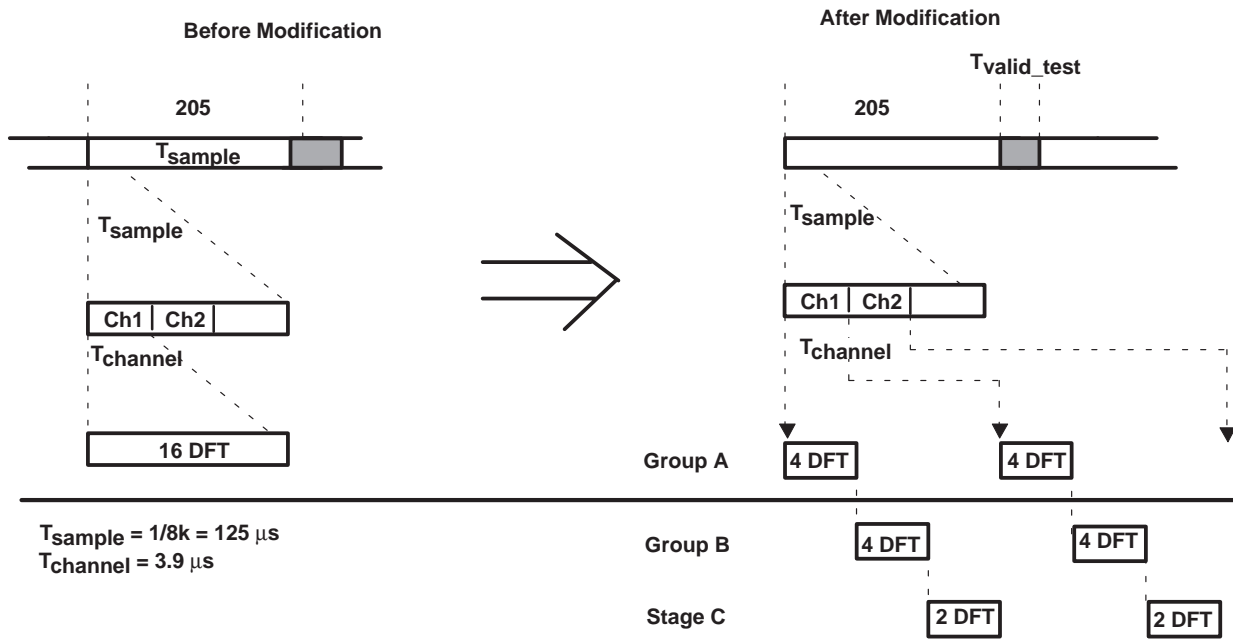
Before the application uses the 'C5x, the device performance is evaluated to determine if it can process 32 channels simultaneously. Figure 4 shows multi-channel operation. The calculations use the input data stream to determine  $T_{\text{sample}}$ ,  $T_{\text{channel}}$ , and  $N$ .  $T_{\text{sample}}$  is the time between interrupts or the sample period.  $T_{\text{channel}}$  is the time to calculate the sample within a channel, and  $N$  is the number of possible channels. The equations for  $T_{\text{sample}}$  and  $N$  are:

$$T_{\text{sample}} = 1/8 \text{ Khz} \tag{4}$$

$$N = T_{\text{sample}}/T_{\text{channel}} \tag{5}$$

For example, if  $T_{\text{sample}} = 125 \mu\text{s}$  and  $N = 32$ , then solving for  $T_{\text{channel}} = 3.9 \mu\text{s}$ . The 'C5x takes 156 cycles with an instruction rate of 50 MIPS (20 ns).

These calculations show that the 'C5x allows each channel 3.9 s to execute its frequency detection program. During this interval, the 'C5x performs 16 DFTs and validates the signal. Since the time period is so small, this poses a performance limitation on the system. A solution to the performance problem is multi-processors. Multi-processors increase the product cost and work against the goal for an inexpensive telecommunications product.



**Note 1:** It takes approximately 195 cycles running at 100 MHz DSP (20 ns) to process 16 DFTs

**Note 2:** 16 DFTs include:

- 697,        697 × 2
- 770,        770 × 2
- 852,        852 × 2
- 941,        941 × 2
- 1209,      1209 × 2
- 1336,      1336 × 2
- 1477,      1477 × 2
- 1633,      1633 × 2

**Note 1:** It takes approximately 195 cycles running at 100 MHz DSP (20 ns) to process 10 DFTs or 4 DFTs if signal does not pass Group A.

**Group A:** 4 DFTs include:

- 1209
  - 1336
  - 1477
  - 1633
- Group B:** 4 DFTs include:
- 697
  - 770
  - 852
  - 941

**Figure 4. Multi-Channel Decoding Timing**

## IMPROVEMENTS TO DTMF ENCODING

Three methods improve the performance of multi-channel DTMF decoders: optimizing the receive interrupt routine, decreasing the number of DFTs, and using a buffer to smooth processing time. Figure 5 shows a modified tone decoder and Figure 6 outlines the program flow for the decoder. Applying all three methods using the modified program flow of Figure 6 improves the performance of DTMF decoders.

### Optimizing the Receive Interrupt Routine

The receive interrupt routine processes two channels in a single interrupt. This combination of channel processing saves one interrupt latency period. The input data uses a  $\mu$ -law-to-linear-conversion look-up table<sup>†</sup>. This table allows the DSP to get a real value for the input sample of the input data.

The routine scales the input sample to eight bits. Two channels of input are read from the 16-bit data receive register (DRR); one from the high byte and one from the low byte.

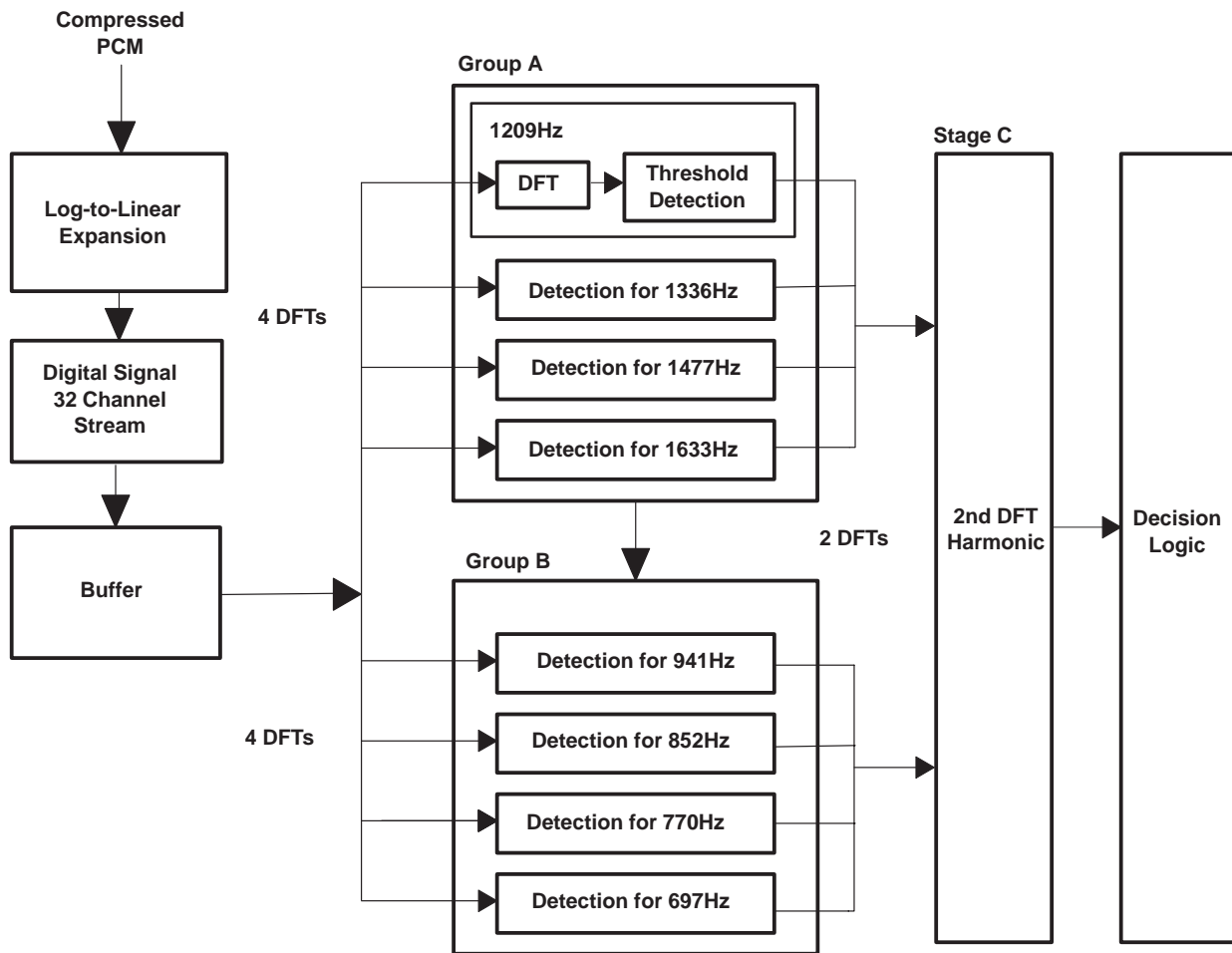


Figure 5. Modified DTMF Tone Decoder

<sup>†</sup>  $\mu$ -law is a companding scheme used in public telephone networks of the US.

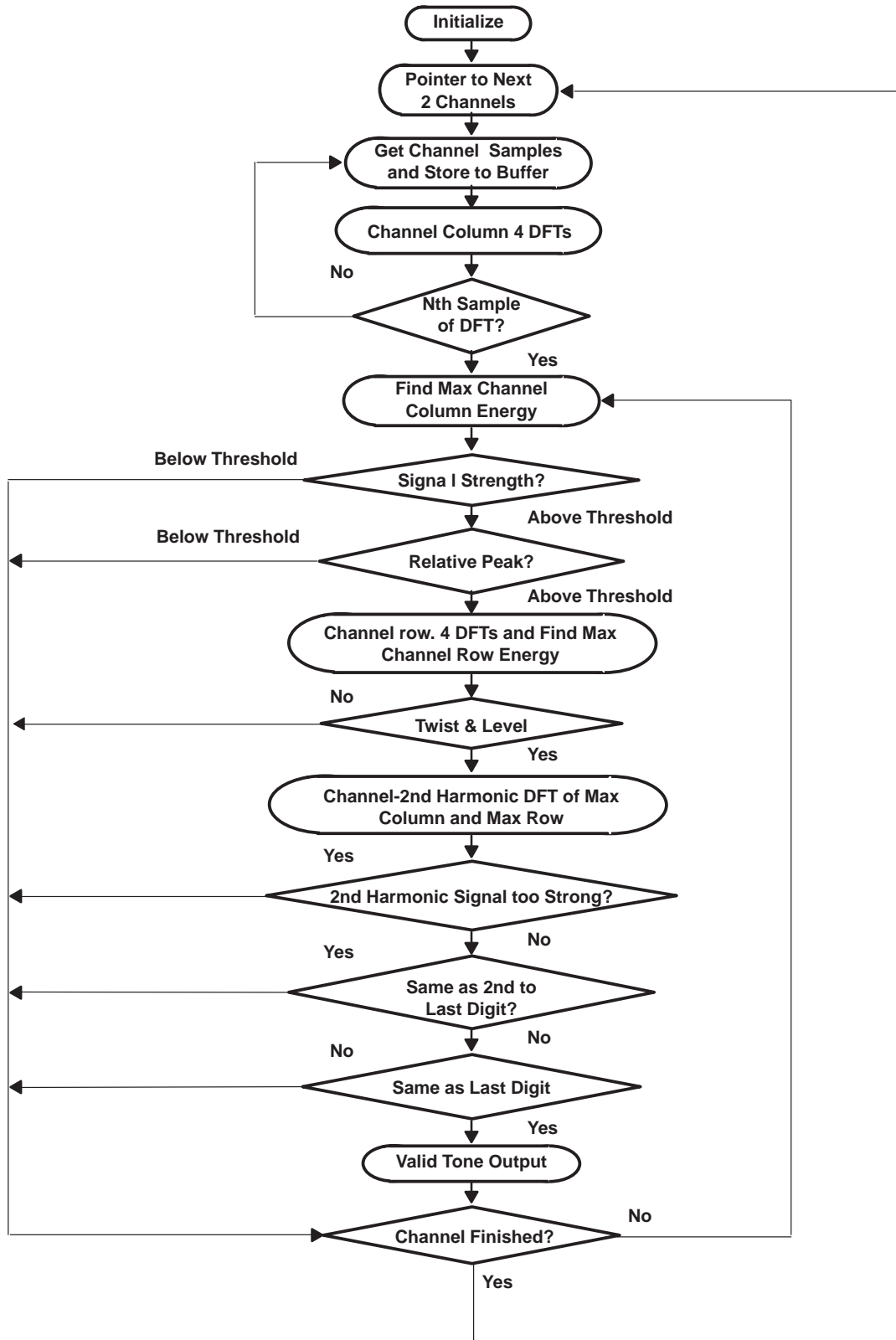


Figure 6. Modified DTMF Tone Detection

## Decreasing the Number of DFTs

The modified tone decoder of Figure 5 shows a process reduction from 16 DFTs to 10 DFTs. This reduction saves processing time in multi-channel operation.

The 10 DFTs are separated into group A, group B and stage C. Group A detects the 4 DFTs belonging to the high frequencies column shown in Figure 1. Group B processes the low frequencies row. Stage C detects the second harmonics and rejects speech from the incoming DTMF tones.

Signal samples go through group A, group B, and stage C (in that order) to detect valid tones. Figure 4 shows the modified processing method.

Between group A and group B processing, the program tests for signal strength and relative peak. The processor squares each DTF value to calculate the magnitude of the signal strength and the maximum value of the four DFTs.

Between group B and stage C processing, the program tests for the twist ratio of the row to the column tone amplitude. The ratio ranges between values with a twist greater than 4 dB and a reverse twist greater than 8 dB<sup>†</sup>.

## Using a Buffer

If all 32 channels are in use the above method of DTF processing does not save time. The method also does not ensure that all 32 channels are processed without data loss. A buffer solves this problem. The buffer stores the input stream that feeds the DTF processing. Since each channel is not constantly in use, the buffer provides the DSP with a greater tolerance for DTF processing and smooths task flow. Figure 7 shows the normalization for multi-channel DTMF.

The buffer size is based upon an AT&T specification that states that the maximum data rate for touch-tone signals is 100 ms/digit. A sample-store interval takes approximately 25 ms. The buffer uses four levels to store the input data during a 100 ms interval. Equation 6 calculates buffer size:

$$\text{Buffer Size} = N \text{ 4 levels} = 32 \text{ K words} \tag{6}$$

where the number of samples, N, is 200-256.

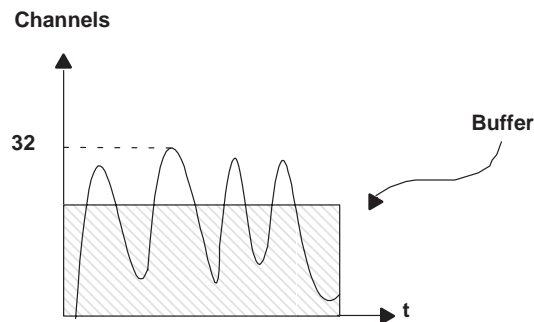


Figure 7. Normalization of the Distribution for Multi-Channel DTMF

<sup>†</sup> AT&T recommends more than 4 dB of twist or 8 dB of reverse twist.

## SUMMARY

The DFT algorithm is compact and needs approximately 200 time-ordered sample lengths to calculate magnitude. The DTMF decoder source code is in Appendix A. The code uses the 10 DFT optimization and buffering to provide 32-channel DTMF processing.

The code in Appendix A needs modification to improve the performance of the DTMF processing. The receive interrupt routine is capable of processing two channels versus the one it now handles. Optimization of the DFTs reduces processing time. A buffer increases performance. All three of these modifications work toward 32-channel DTMF

All of the improvements achieve the goal processing 32 channels in a single TMS320C5x DSP. Other DSP functions, such as DTMF encoding, voice compression/expansion, and voice mail require either a multi-processor or a high performance DSP.

## REFERENCES

1. Texas Instruments, Digital Signal Processing Applications with the TMS320 Family, Theory, Algorithms, and Implementations Volume 1, Topic 19, "Add DTMF Generation and Decoding to DSP- $\mu$ P Design" (1989).

## APPENDIX A—DTMF DECODE PROGRAM

```

;*****
;*TMS320C5x DTMF Tone decoder source code example
;*****
; Column
;
;           1       2       3       4
;           1209hz 1336hz 1477hz 1633hz
;
;           +-----+
;           | 1  _| 2  _| 3  _| A  _|
;           | 697hz | 0  | 1  | 2  | 3  |
;
;           +-----+-----+-----+-----+
;           | 4  _| 5  _| 6  _| B  _|
;           | R 770hz | 4  | 5  | 6  | 7  |
;           | O     | +-----+-----+-----+
;           | W 3   | | 7  _| 8  _| 9  _| C  _|
;           | 852hz | | 8  | 9  | a  | b  |
;
;           +-----+-----+-----+-----+
;           | 4   | | *  _| 0  _| #  _| D  _|
;           | 941hz | | |c|  |d|  |e|  |f|
;
;           +-----+
;
;----- DTMF decoder -----
;Y(n) = X(n) + 2Cos(coef)*Y(n-1) - Y(n-2) + 2Cos(coef)*Y(n-1)
; k = N * Freq / Sample rate
;----- Coefficient -----
;coefficent = 32768 * Cos(360 * Freq / Sample rate)
;coef = 360 * Freq / Sample rate
;Sample rate = 8kHz
;----- Data Segment -----
.data
Table .set $
Baselr .word 32768 * 8538/10000 ;27906: 697Hz
        .word 32768 * 8226/10000 ;26802: 770Hz
        .word 32768 * 7843/10000 ;25597: 851Hz
        .word 32768 * 7391/10000 ;24295: 941Hz
        .word 32768 * 5821/10000 ;19057: 1209Hz
        .word 32768 * 4982/10000 ;16527: 1336Hz
        .word 32768 * 3993/10000 ;12945: 1477Hz
        .word 32768 * 2843/10000 ; 9166: 1633Hz
;2nd harmonic rate coefficient
Harm1r .word 32768 * 4582/10000 ; 15036:2 x 697Hz
        .word 32768 * 3535/10000 ; 11287:2 x 770Hz
        .word 32768 * 2304/10000 ; 7363:2 x 851Hz
        .word 32768 * 925/10000 ; 3323:2 x 941Hz
        .word 32768 * -3224/10000 ;-10805:2 x 1209Hz

```

```

.word 32768 *-5036/10000      ; -16384:2 x 1336Hz
.word 32768 *-6811/10000     ; -22153:2 x 1477Hz
.word 32768 *-8384/10000     ; -27440:2 x 1633Hz
;   word 27906,26802,25597,24295, 19057, 16527, 12945, 9166
;   word 15036,11287, 7363, 3323,-10805,-16384,-22153,-27440
.word 1,205,0,0,0,0,0,0,0,B1QueCh0
.word B1F11Ch31,PProcFlg0,B1QueCh31,B1F31Ch31,0,PProcFlg0,3
.word 8000h,2,3,4,0,204,200,06h,16h,B1QueCh0
Len.set $ - Table
;--- u_law table ---
uTab .word 0e0a1h,0e1a1h,0e2a1h,0e3a1h,0e4a1h,0e5a1h,0e6a1h,0e7a1h
      .word 0e8a1h,0e9a1h,0eaa1h,0ebal1h,0ecal1h,0eda1h,0eea1h,0efal1h
      .word 0f061h,0f0e1h,0f161h,0f1e1h,0f261h,0f2e1h,0f361h,0f3e1h
      .word 0f461h,0f4e1h,0f561h,0f5e1h,0f661h,0f6e1h,0f761h,0f7e1h
      .word 0f841h,0f881h,0f8c1h
      .word 0f901h,0f941h,0f981h,0f9c1h,0fa01h,0fa41h,0fa81h,0fac1h
      .word 0fb01h,0fb41h,0fb81h,0fbc1h
      .word 0fc01h,0fc31h,0fc51h,0fc71h,0fc91h,0fcb1h,0fcd1h,0fcf1h
      .word 0fd11h,0fd31h,0fd51h,0fd71h,0fd91h,0fdb1h,0fdd1h,0fdf1h
      .word 0fe11h,0fe29h,0fe39h,0fe49h,0fe59h,0fe69h,0fe79h,0fe89h
      .word 0fe91h,0fea9h,0feb9h,0fec9h,0fed9h,0fee9h,0fef9h
      .word 0ff09h,0ff19h,0ff25h,0ff2dh,0ff35h,0ff3dh,0ff45h,0ff4dh
      .word 0ff55h,0ff5dh,0ff65h,0ff6dh,0ff75h,0ff7dh,0ff85h,0ff8dh
      .word 0ff95h,0ff9dh
      .word 0ffa3h,0ffa7h,0ffabh,0ffafh,0ffb3h,0ffb7h,0ffbbh,0ffbfbh
      .word 0ffc3h,0ffc7h,0fccbh,0fccfh,0ffd3h,0ffd7h,0ffdbb,0ffdfh
      .word 0ffe2h,0ffe4h,0ffe6h,0ffe8h,0ffeah,0ffech,0ffeeh
      .word 0fff0h,0fff2h,0fff4h,0fff6h,0fff8h,0fffah,0fffch,0fffeh
      .word 00000h
;negative
.word 01f5fh,01e5fh,01d5fh,01c5fh,01b5fh,01a5fh,0195fh,0185fh
.word 0175fh,0165fh,0155fh,0145fh,0135fh,0125fh,0115fh,0105fh
.word 00f9fh,00f1fh,00e9fh,00e1fh,00d9fh,00d1fh,00c9fh,00c1fh
.word 00b9fh,00b1fh,00a9fh,00a1fh,0099fh,0091fh,0089fh,0081fh
.word 007bfh,0077fh,0073fh
.word 006ffh,006bfh,0067fh,0063fh
.word 005ffh,005bfh,0057fh,0053fh,004ffh,004bfh,0047fh,0043fh
.word 003ffh,003cfh,003afh,0038fh,0036fh,0034fh,0032fh,0030fh
.word 002efh,002cfh,002afh,0028fh,0026fh,0024fh,0022fh,0020fh
.word 001efh,001d7h,001c7h,001b7h,001a7h,00197h,00187h,00177h
.word 0016fh,00157h,00147h,00137h,00127h,00117h,00107h
.word 000f7h,000e7h
.word 000dbb,000d3h,000cbh,000c3h,000bbh,000b3h,000abh,000a3h
.word 0009bh,00093h,0008bh,00083h,0007bh,00073h,0006bh,00063h
.word 0005dh,00059h,00055h,00051h,0004dh,00049h,00045h,00041h

```



```

        .word    0003dh,00039h,00035h,00031h,0002dh,00029h,00025h,00021h
        .word    0001eh,0001ch,0001ah,00018h,00016h,00014h,00012h,00010h
        .word    0000eh,0000ch,0000ah,00008h,00006h,00004h,00002h,00000h
;----- bss Segment -----
        .bss    B1QueCh0,205*32 0      ;Queue Buffer Block1 205*32
B1QueCh31 .set    B1QueCh0+205*31
        .bss    B2QueCh0,205*32      ;Queue Buffer Block2 205*32
B2QueCh1  .set    B2QueCh0+205
        .bss    B3QueCh0,205*32      ;Queue Buffer Block3 205*32
B3QueCh1  .set    B3QueCh0+205
        .bss    B4QueCh0,205*32      ;Queue Buffer Block4 205*32
B4QueCh1  .set    B4QueCh0+205
        .bss    B1F11Ch31,6*32      ;Filter result Block1 6*32
B1F31Ch31 .set    B1F11Ch31+4
B1F11Ch30 .set    B1F11Ch31+6
        .bss    B2F11Ch31,6*32      ;Filter result Block2 6*32
B2F11Ch30 .set    B2F11Ch31+6
        .bss    B3F11Ch31,6*32      ;Filter result Block3 6*32
B3F11Ch30 .set    B3F11Ch31+6
        .bss    B4F11Ch31,6*32      ;Filter result Block4 6*32
B4F11Ch30 .set    B4F11Ch31+6
;----- bK0 & bK1 =100h-2ffh -----
CoefBase1r .usect "varity",512
CoefBase4r .set    CoefBase1r+3
CoefBase3c .set    CoefBase1r+6
CoefHarm1r .set    CoefBase1r+8
CoefHarm1c .set    CoefBase1r+12
CoefHarm4c .set    CoefBase1r+15
One        .set    CoefBase1r+16      ;*1
FilCout    .set    CoefBase1r+17      ;*receive code counter
Cout0Flg   .set    CoefBase1r+18      ;*count 0 first time=0->-1 2's =;-1->0
PQueFlg    .set    CoefBase1r+19      ;*Recv int use Nth queue block buffer
PProcFlg0  .set    CoefBase1r+20      ;*Queue buffer blockN receive ok = -1
PProcFlg1  .set    CoefBase1r+21
PProcFlg2  .set    CoefBase1r+22
PProcFlg3  .set    CoefBase1r+23
DtmfFlag0  .set    CoefBase1r+24      ;*DTMF On/Off flag 16*2 = 32ch
DtmfFlag1  .set    CoefBase1r+25
PQue       .set    CoefBase1r+26      ;*Recv int use Queue pointer
PB1F11Ch31 .set    CoefBase1r+27      ;*pointer B1F11Ch31
PPProcFlg0 .set    CoefBase1r+28      ;*pointer PProcFlg0
PB1QueCh31 .set    CoefBase1r+29      ;*pointer B1QueCh31
PB1F31Ch31 .set    CoefBase1r+30      ;*pointer B1F31Ch31
PGropSave0 .set    CoefBase1r+31      ;*Main program ar0 save
PGropSave1 .set    CoefBase1r+32      ;*Main program ar1 save

```

```

PGropSave2 .set CoefBaselr+33 ;*Main program ar2 save
NegMax .set CoefBaselr+34 ;*8000
Two .set CoefBaselr+35 ;*2
Three .set CoefBaselr+36 ;*3
Four .set CoefBaselr+37 ;*4
PChSave .set CoefBaselr+38 ;*Main program ch process save
BaseLen .set CoefBaselr+39 ;*204(205)
HarmLen .set CoefBaselr+40 ;*200(201)
D06h .set CoefBaselr+41 ;*06h
D16h .set CoefBaselr+42 ;*16h
PB1QueCh0 .set CoefBaselr+43 ;*pointer B1QueCh0
;----- Not Initialization -----
FreqCout4 .set CoefBaselr+50
TempInt .set CoefBaselr+51
TempInt1 .set CoefBaselr+52
TempMain .set CoefBaselr+53
ColMaxNo .set CoefBaselr+54
RowMaxNo .set CoefBaselr+55
ColMaxDat .set CoefBaselr+56
RowMaxDat .set CoefBaselr+57
test .set CoefBaselr+58
test1 .set CoefBaselr+59
FilR11b .set CoefBaselr+60 ;FilR11b-FilR42b =4*2=8
FilR12b .set CoefBaselr+61
FilR31b .set CoefBaselr+64
FilR32b .set CoefBaselr+65
FilR41b .set CoefBaselr+66
FilR42b .set CoefBaselr+67
PFilC11b .set CoefBaselr+68 ;FilC11b-FilC32b =3*2=6
PFilC22b .set CoefBaselr+71
PFilC31b .set CoefBaselr+72
PFilC32b .set CoefBaselr+73
PQueBuf .set CoefBaselr+74 ;Pointer Queue (Main program)
FilR01h .set CoefBaselr+75 ;FilC01h-Filc02h =2*2=4
FilR02h .set CoefBaselr+76
FilC01h .set CoefBaselr+77
FilC02h .set CoefBaselr+78
Last0Ch0 .set CoefBaselr+79
Last1Ch0 .set CoefBaselr+80 ;Last0Ch0-Last2Ch31 = 3*32=96
Last2Ch0 .set CoefBaselr+81
Last2Ch31 .set CoefBaselr+174
F11Ch31 .set CoefBaselr+175 ;F11Ch31-F32ch0 = 6*32=192
F12Ch31 .set CoefBaselr+176
F21Ch31 .set CoefBaselr+177
F32Ch0 .set CoefBaselr+366

```

```

;----- Set up the ISR vector -----
    .sect "vectors"
    b    Main
    b    Int1          ;02  INT1 interrupt
    b    Int2          ;04  INT2 interrupt
    b    Int3          ;06  INT3 interrupt
    .space 2 * 16      ;08  TINT interrupt
Rint:  b    RecvInt    ;0A  Serial port receive interrupt RI
XInt:  b    TransInt   ;0C  Serial port transmit interrupt X
TRnt:  b    TRecvInt   ;0E  TDM receive interrupt
TXnt:  b    TTransInt  ;10  TDM transmit interrupt
    .space 18 * 16    ;12

;-----

    .text
    .mmregs
;*****
;
;               Main program
;*****
;----- Initialization -----
Main:
    setc  intm          ;ST0
    setc  ovm
    ldp   #0
    opl   #003ch,PMST   ;ndx=1 ram=ovly=1
    splk  #00c8h,SPC    ;soft=free=0 rrst=xrst=1 txm=0 mcm=0 fsm=1 fo=0
    splk  #00c8h,TSPC   ;same SPC
    spm   0             ;ST1
    setc  sxm
    clrc  cnf           ;B1 = data mem
    lmmr  INDX,#Two     ;indx=2
    zap
    sacl  CWSR          ;CWSR=PDWSR=IOWSR=0
    sacl  PDWSR
    sacl  IOWSR
    splk  #06h,IMR      ;Series Recv Int on & Int2 & Int3
;----- block move -----
    mar   *,ar1         ;table move
    lar   ar1,#CoefBaselr
    rpt   #Len-1
    blpd  #Table,*+
;-----

    lar   ar1,#F11Ch31  ;clear filter result buffer 6*32
    rpt   #191
    sacl  *+
    clrc  intm

```

```

        ldp    #CoefBase1r          ;check if need to process
MainChk0:
        lar    ar0,PGropSave0
        lar    ar1,PGropSave1
        lar    ar2,PGropSave2
MainChk1
        lacc   *
        bcnd  MainProc0,neq        ;PProcFlg(n)=? -1
        mar   **+,ar0
        mar   **+,ar2
        banz  MainChk1,*-,ar1
        lar   ar0,#0
        lar   ar1,#PProcFlg0
        lar   ar2,#3
        b     MainChk1
MainProc0:
        zap
        sacl  *                    ;0 -> PProcFlg(n)
        sar   ar0,PGropSave0
        sar   ar1,PGropSave1
        sar   ar2,PGropSave2
        lacc  #31                  ;set PChSave=#31(32ch)
        sacl  PChSave
MainProc1
        lt    PGropSave0          ;Group start then set
        mpy   #192                ;1.PFilC31b,PFilC31b
        pac
        adds  PB1F31Ch31         ;PB1F31Ch31
        sacl  PFilC31b
        add   One
        sacl  PFilC32b
        lt    PGropSave0          ;2.PB1QueCh31
        mpy   #205*32
        pac
        bd    MainProc2
        adds  PB1QueCh31         ;PB1QueCh31
        sacl  PQueBuf
MainProc3:
        sar   ar1,PChSave         ;channel change then
        lacc  PQueBuf,0           ;1.PQueBuf-205,channel-1
        sub   #205
        sacl  PQueBuf
        lacc  PFilC31b           ;2.PFilC31b+6&PFilC32b+6
        add   #6
        sacl  PFilC31b

```

```

        add    One
        sacl  PFilC32b
MainProc2:
        lmmr  INDX,#Two          ;indx=2
;*****
;    Caculate Energy at Col 1,2,3,4 and
;    Find Max Energy of Col 1,2,3,4
;*****
;ENERGY = [(Y(n-2)-Y(n-1))/2]^2 - Y(n-1)*Y(n-2)*(coef-1)
EnergyCol:
        lar   ar0,Two
        lar   ar1,PFilC32b
        lar   ar2,#CoefBase3c
EnergyCol0:
        call  EnergySr,*,ar2
        sach  *,1,ar0
        banz  EnergyCol0,*,ar1
;----- Find Col1,2,3 Energy Max -----
;ColMaxDat = col peak data ColMaxNo= col peak number
FindColMax:
;    mar    *,ar1
        lacc  Two          ;ColMaxNo=2
        sacl  ColMaxNo
FindCol
        lar   ar2,One          ;ar2=1
        lar   ar1,PFilC31b     ;ColMaxDat=(PFilC31b) & ar1=PFilC21b
        lacc  *0-
        sacl  ColMaxDat
FindCol0
        lacc  ColMaxDat
        sub   *
        bcnd  FindCol1,geq     ;col_max - (ar1)
        sar   ar2,ColMaxNo
        lacc  *
        sacl  ColMaxDat
FindCol1
        mar   *0-,ar2
        banz  FindCol0,*,ar1
;*****
;*    Check Col 1,2,3,4 level and Relative peak
;*****
ChkLev:
        lacc  ColMaxDat          ;ColMaxDat must >2^4
        sub   One,4
        bcnd  NotCode,lt

```

```

;----- Check Col1,2,3 Relative ratio -----
ChkRelCol:
;   mar   *,ar1
   lar   ar0,Two           ;ar0=2
   lar   ar1,PFilC31b     ;ar1=PFilC31b
   lar   ar2,One          ;ar2=1
   lt    ColMaxDat        ;P=ColMaxDat*683
   mpy   #683
ChkRelCol0
   lacc  *0-,12,ar0       ;(ar1) - ColMaxDat*683 & ar1-2
   spac
   bcnd  ChkRelCol1,lt
   mar   *,ar2            ;ar2 - 1
   mar   *-,ar0
ChkRelCol1
   banz  ChkRelCol0,*-,ar1 ;ar0 =?0
   mar   *,ar2
   banz  NotCode,*,ar1    ;ar2=?0
;----- Clear ROW & Harmonic Filter result -----
ClearFil:
   lar   ar1,#FilR11b
   zap
   rpt   #7
   sacl  *+
   lar   ar1,#FilR01h
   rpt   #3
   sacl  *+
;*****
;   Process DFT of Row 1,2,3,4
;*****
DFTRow:
   lar   ar0,BaseLen
   lar   ar1,#FilR12b
   lar   ar2,PQueBuf
   lar   ar3,#CoefBaselr
   lacc  Four
   sacl  FreqCout4
   call  DFTCall,*,ar3
;*****
;   Caculate Energy at Row 1,2,3,4 and
;   Find Max Energy of Row 1,2,3,4
;*****
;ENERGY = [(Y(n-2)-Y(n-1))/2]^2 - Y(n-1)*Y(n-2)*(coef-1)
EnergyRow:
   lar   ar0,Three

```

```

        lar    ar1,#FilR42b
        lar    ar2,#CoefBase4r
EnergyRow0:
        call   EnergySr,*,ar2          ;21
        sach   *-,1,ar0                ;23..ar1 - 1 (Y(n-2)
        banz   EnergyRow0,*-,ar1       ;24..ar0(FreqCount)=?0 ar0-1
;----- Find Row 1,2,3,4 Energy Max -----
;RowMaxDat = row peak data    RowMaxNo= row peak number
FindRowMax:
;    mar    *,ar1
        lacc   Three                    ;RowMaxNo=3
        sacl   RowMaxNo
FindRow
        lar    ar2,Two                  ;ar2=2
        lar    ar1,#FilR41b            ;RowMaxDat=(#FilR41b) & ar1=#FilR31b
        lacc   *0-
        sacl   RowMaxDat
FindRow0
        lacc   RowMaxDat
        sub    *
        bcnd   FindRow1,geq            ;RowMax - (ar1)
        sar    ar2,RowMaxNo
        lacc   *
        sacl   RowMaxDat
FindRow1
        mar    *0-,ar2
        banz   FindRow0,*-,ar1
;*****
;    Check Peak level and Twist
;*****
TwistLevel:
        lacc   ColMaxDat
        sub    RowMaxDat
        bcnd   TwistLevel0,geq
TwistRev
        lac    ColMaxDat                ;row > col
        sub    One,4
        bcnd   NotCode,lt
        lac    RowMaxDat                ;row_max - col_max*12
        sub    ColMaxDat,3              ;8dB = 6
        sub    ColMaxDat,2
        bcnd   NotCode,geq
        b      DFTHarmRow
TwistLevel0:
        lac    RowMaxDat                ;col > row

```

```

sub    One,4
bcnd  NotCode,lt
lac   ColMaxDat           ;col_max - row_max*3
sub   RowMaxDat           ;4dB = 3
sub   RowMaxDat,1
bcnd  NotCode,geq
;*****
;    Process 2nd Harmonic DFT of Max Row Energy and
;    Check Row Harmonic Energy
;*****
DFTHarmRow:
    mar    *,ar3
    lmmr   INDX,#RowMaxNo
    lar    ar0,HarmLen
    lar    ar1,#FilR02h
    lar    ar2,PQueBuf
    lar    ar3,#CoefHarmlr
    mar    *0+,ar3
    lt     *,ar2           ;1..T = Cos(*)
DFTHarmRow0
    lacc   +,12,ar1       ;3..X(n)
    sub    -,16           ;X(n)-Y(n-2)  --> Acc
    mpy    *              ;Cos(*) * Y(n-1)  --> P
    ltd    *,ar3         ;X(n)+ Cos(*) * Y(n-1)-Y(n-2)Y(n-1)-->Y(n-2)
    lta    *,ar1         ;X(n)+2Cos(*) * Y(n-1)-Y(n-2)
    apac
    apac
    sach   +,0,ar0       ;2..--> Y(n-1) Q30
    bcnd   HarmRowLev,ov  ; check overflow
    banz   DFTHarmRow0,*,ar2 ;4..ar0(len)=?0
;----- Check Row Harmonic Energy -----
HarmRowLev:
;    mar    *,ar2
    lar    ar2,#CoefHarmlr ;ar2 = harmonic coefficient pointer
    mar    *0+,ar2         ;= #coeff + row_max*2
    lar    ar1,#FilR02h   ;ar1 = harmonic data pointer= #dat + row_mxp*2
    call   EnergySr,*,ar2
    lt     RowMaxDat      ;harmonic energy row_max/4
    mpy    #4096
    spac
    spac
    bcnd   NotCode,geq

```



```

;*****
;      Process 2nd Harmonic DFT of Max Col Energy and
;      Check Col Harmonic Energy
;*****
DFTHarmCol:
    mar    *,ar3
    lmmr   INDX,#ColMaxNo
    lar    ar0,HarmLen
    lar    ar1,#FilC02h
    lar    ar2,PQueBuf
    lar    ar3,#CoefHarm1c
    mar    *0+,ar3
    lt     *,ar2          ;1..T = Cos(*)
DFTHarmCol0
    lacc   *+,12,ar1      ;3..X(n)
    sub    *-,16          ;X(n)-Y(n-2)--> Acc
    mpy    *              ;Cos(*) * Y(n-1)--> P
    ltd    *,ar3          ;X(n)+ Cos(*) * Y(n-1)-Y(n-2)Y(n-1)-->Y(n-2)
    lta    *,ar1          ; X(n)+2Cos(*) * Y(n-1)-Y(n-2)
    apac
    apac
    sach   *+,0,ar0       ;2..--> Y(n-1) Q30
    bcnd   HarmColLev,ov  ;check overflow
    banz   DFTHarmCol0,*-,ar2 ;4..ar0(len)=?0
;----- Check Col Harmonic Energy -----
HarmColLev:
    mar    *,ar2
    lar    ar2,#CoefHarm1c ;ar2 = harmonic coefficient pointer
    mar    *0+,ar2         ;= #Coef + ColMax*2
    lar    ar1,#FilC02h    ;ar1=harmonic data pointer= #dat + ColMax*2
    call   EnergySr,*,ar2
    lt     ColMaxDat       ;harmonic energy ColMax/4
    mpy    #4096
    spac
    spac
    bcnd   NotCode,geq
;*****
;      Check for valid number
;      Compare number with last number and 2nd to last number
;*****
;----- load recognized number -----
    mar    *,ar1
    lt     PChSave         ;ch*3 + #Last1Ch0
    mpy    Three
    lacc   #Last0Ch0

```

```

    apac
    sac1  TempMain,0
    lar   ar1,TempMain
    mar   *+,ar1           ;Last0Ch0+1 = Last1Ch0
    dmov  *-,              ;last1 --> last2
    dmov  *,               ;last0 --> last1
;----- Check for new number -----
ChkNewCode:
    lacc  RowMaxNo,2       ;row_mxp*2^2 + col_mxp + Ch*2^8--> last0
    add   ColMaxNo
    add   PChSave,8
    sac1  *+               ;--> last0
    mar   *+,ar1
    sub   *-,              ;last0 =? last2
    bcnd  NotCode1,eq
    lac   *-,              ;last0 =? last1
    sub   *,0,ar1
    bcnd  ChkNewCode0,neq
    out   *,3              ;Output Last0 Digital data
ChkNewCode0
    lar   ar1,PChSave
    banz  MainProc3,*-,ar1 ;PChSave=?0 & PChSave-1
    b     MainChk0
;----- NOT New CODE -----
NotCode:
    lt    PChSave          ;ch*3 + #Last0Ch0
    mpy   Three
    lacc  #Last0Ch0
    apac
    sac1  TempMain,0
NotCode1
    lar   ar1,TempMain
    mar   *,ar1
    lacc  #0ffh
    bd    ChkNewCode0
    dmov  *,               ;last0 --> last1
    sac1  *
;*****
;          DFT Subroutine
;*****
DFTCall:
    lt    *,ar2            ;1..T = Cos(*)
DFTCall0
    lacc  *+,12,ar1        ;3..X(n)
    sub   *-,16            ;X(n)-Y(n-2)--> Acc

```

```

    mpy    *                ;Cos(*) * Y(n-1) --> P
    ltd    *,ar3            ;X(n)+ Cos(*) * Y(n-1)-Y(n-2)Y(n-1)-->Y(n-2)
    lta    *,ar1            ;X(n)+2Cos(*) * Y(n-1)-Y(n-2)
    apac
    apac                ;X(n)+2*2Cos(*)*Y(n-1)-Y(n-2)
    sach    *+,0,ar0        ;2.--> Y(n-1) Q30
    bcnd   DFTCall1,ov      ;check overflow
    banz   DFTCall0,*-,ar2  ;4..ar0(len)=?0
DFTCall1
    lar    ar0,BaseLen
    lar    ar2,PQueBuf
    mar    *,ar1
    mar    *0+,ar3
    mar    *+,ar3
    lacc   FreqCout4
    sub    One
    sac1   FreqCout4
    bcnd   DFTCall,neq
    ret

;*****
;          ENERGY Subroutine
;*****
EnergySr:
    lac    NegMax,15        ;(coef-1)/2
    add    *-,15,ar1        ;ar2=&coeff-1
    sach   TempMain
    lt     *-                ;Y(n-2)*(coef-1)/2
    mpy    TempMain
    pac
    sach   TempMain,1
    lt     *+                ;Y(n-1)*Y(n-2)*(coef-1)/2
    mpy    TempMain
    pac
    sach   TempMain,1
    lac    *-,15            ;ABS[Y(n-2)-Y(n-1)]/2
    sub    *,15
    abs
    sach   *
    lt     *                ;[Y(n-2)-Y(n-1)]/2 ^2
    mpy    *
    pac                ;[(Y(n-2)-Y(n-1))/2]^2-Y(n-1)*Y(n-2)*(coef-1)
    sub    TempMain,15
    ret

```

```

;*****
;
;           Receive int subroutine
;           Processing 2 channels in single Interrupt
;*****
RecvInt:
    ldp    #TempInt
    lamm   DRR           ;DRR u_law Expanding
    sacl   TempInt
    mar    *,ar4
    and    #0ffh
    add    #uTab
    tblr   *,ar5
    lar    ar7,#3       ;4 Freq of 1st channel
RecvInt4
    lt     *-,ar4       ;4 frequency IIR
    lacc   *,12,ar6
    sub    *-,16
    mpy    *
    ltd    *
    apac
    apac
    apac
    sach   *-,0,ar7
    bcnd   RecvInt5,ov
    banz   RecvInt4,*-,ar5
RecvInt5
    adrkl #4           ;ar5+4 pointer process
    lar    ar7,#3
    lacl   TempInt     ;Get high byte data
    bsar   8
    mar    *,ar4
    adrkl #205
    add    #uTab
    tblr   *,ar5
RecvInt6
    lt     *-,ar4       ;4 frequency IIR of 2nd channel
    lacc   *,12,ar6
    sub    *-,16
    mpy    *
    ltd    *
    apac
    apac
    apac
    sach   *-,0,ar7
    bcnd   RecvInt7,ov

```

```

        banz    RecvInt6,*-,ar5
RecvInt7
        adrk   #4                ;pointer process
        mar    *,ar4
        adrk   #205
        lacc   #0ffffh           ;output level 0 to phone
        samm   DXR
        rete

;*****
;   Other Interrupt Subroutines
;*****
;----- TDM transmit ISR -----
TTransInt:
        rete
;----- TDM receive ISR -----
TRecvInt:
        rete
;----- DXR transmit ISR -----
TransInt:
        rete
;----- INT1 ISR -----
Int1:
        rete
;----- INT2 ISR(Frame 0) -----
Int2:
        ldp    #FilCout
        lmmr   IMR,#D16h         ;turn on IMR Recv int
        lacc   FilCout,0         ;FilCout =?0
        bcnd   Int10,neq
        lacc   Cout0Flg         ;Cout0Flg complement
        cml
        sacl   Cout0Flg
        bcnd   Int11,neq
        lamm   DRR               ;read DRR data
        lacc   #205
Int10
        sub    One                ;FilCout - 1
        sacl   FilCout,0
        lar    ar5,#CoefBase3c   ;set Ar5 - Ar6
        lar    ar6,#F32Ch0
        lar    ar4,PQue          ;set ar4 = PQue & PQue + 1
        lacc   PQue
        add    One
        sacl   PQue
        rete

```

```

Int11:
    lmmr    IMR,#D06h           ;stop Recv Int
    lt      PQueFlg             ;caculate Filter result Block addr
    mpy     #192
    pac
    adds    PB1F11Ch31
    sacl    TempInt
    mar     *,ar5               ;Block move F11Ch31 --> BnF11Ch31
    lar     ar5,TempInt
    rpt     #191
    bldd    #F11Ch31,*+
    lar     ar5,#F11Ch31       ;Clear F11ch31 - F32Ch0
    zap

rpt #191
    sacl    *+
    lacc    PQueFlg             ;set PProcFlg(n) = -1
    adds    PPProcFlg0
    sacl    TempInt
    lar     ar5,TempInt
    zap
    sub     One
    sacl    *
    lacc    PQueFlg             ;PQueFlg+1 & #3 -->PQueFlg
    add     One
    and     Three
    sacl    PQueFlg
    lt      PQueFlg             ;caculus PQue addr
    mpy     #205*32
    pac
    adds    PB1QueCh0
    sacl    PQue
    rete

;----- INT3 ISR(Receive Code) -----
;Get Tone Format: High 8Bit          Low 8bit
; Bit0-4 = channel   Bit0 = 0:DTMF On 1:DTMF Off
Int3:
    ldp     #TempInt
    in      TempInt,PA0         ;in data
    bit     TempInt,15
    lacc    TempInt,8           ;Get channel
    and     #0fh,16
    sach    TempInt1
    bcddd   Int21,tc
    lt      TempInt1           ;mark or unmark DtmfFlag bit
    lact    One

```

```
        bit    TempInt,11
        cpl
        bcnd  Int22,tc
        and   DtmfFlag0
        sac1  DtmfFlag0
        rete

Int22:
        and   DtmfFlag1
        sach  DtmfFlag1
        rete

Int21:
        bit    TempInt,11
        bcnd  Int22,tc
        or    DtmfFlag0
        sac1  DtmfFlag0
        rete

Int23:
        or    DtmfFlag1
        sach  DtmfFlag1
        rete
        .end
```