

Apple II Digital Storage Oscilloscope

Use minimal hardware, Applesoft BASIC, and 6502 machine language to convert an Apple II into a triggered digital "storage oscilloscope."

This article gives complete details on building a peripheral card that fits in slot #7 of the Apple II computer. The card has an audio frequency buffer and attenuation circuit, plus a very easy-to-use analog-to-digital converter which is capable of sampling the signal every 25 microseconds. The software is a combination program of Applesoft BASIC and 6502 machine language. It prompts you to initiate an audio input, then waits for it. It has a trigger threshold. When the signal level exceeds the threshold, the circuit takes 19,600 samples spaced about 40 μ S apart, storing each one in upper RAM (starting at \$7370 in a 48K space). Then a high-resolution plot of the first 280 samples — one screen — is displayed.

The caption beneath the screen-plot gives start and stop times of that screen in milliseconds. It also prompts you to push certain keys to look at other screen-plots. There are 70 screen-plots altogether, each covering roughly 12 milliseconds. Thus, the original signal is sampled for about .84 seconds. See figure 1 for an example of a screen-plot. The circuitry and code are fully operational, but the information here should really be taken as a suggestive guide for customizing the ideas to your own needs.

Physically, the circuit is connected together by point-to-point soldering of wire-wrap type wire on a California Computer Systems model 7500 proto-type board (see figure 2). (This board costs \$21, and the labels on both sides

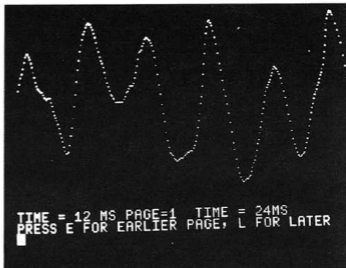


Figure 1: Sample plot of waveform produced by a conga drum.

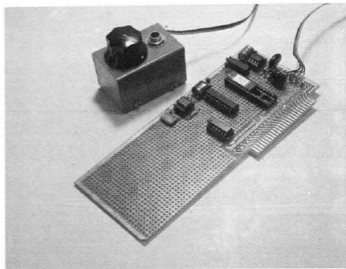


Figure 2: Component side of the ADC board.

for all edge-connector terminals are very helpful for wiring and checkout.) There is a two-wire ribbon cable from the card to a little metal box with a phone jack for plugging in the signal, and a potentiometer for attenuating the input signal.

You do well if you find an integrated circuit as friendly as the Analog Devices AD570JD. This costs \$22 (in singles) but is a completely self-contained, successive-approximation, analog-to-digital converter. It has one analog input (choose unipolar 0-10V input by grounding the BIPOLAR CONTROL pin, or leave the pin floating for bipolar +5V input), and 8 digital outputs. There is one input control line called BLANK/CONVERT (B/C), and one output acknowledgement line, DATA READY (DR). In BLANK mode the digital output lines are floated (tri-state, high impedance condition). Nothing

happens until B/C is brought low, thereby switching the unit into CONVERT mode. After grinding out the answer in 25 μ S, the result appears latched onto the eight output lines, and DR is brought low. That is it. Bring the unit back into BLANK, and start over. It has to stay in BLANK a minimum of 2 μ S before another conversion is initiated. Also, in bipolar mode the output is offset binary [zero signal gives output 128]. At this point you can foresee that the software loop for filling RAM area with samples of audio signal will have the following steps:

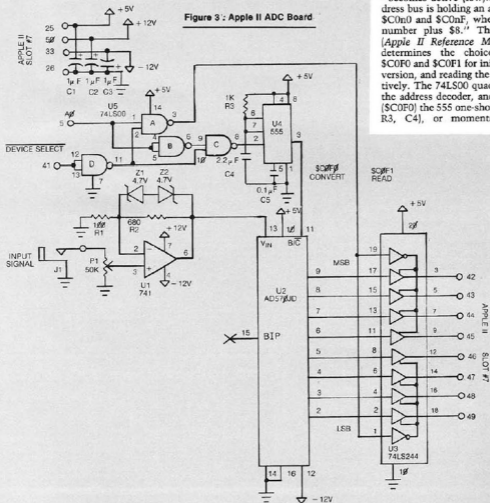
1. initiate a conversion by bringing B/C from low to high for 2 μ S;
2. wait at least 25 μ S, and use the time to check for done and to compute the address of the next location to stick the answer;

3. read the answer;
4. stick it in RAM;
5. go back to 1.

Here is how the circuit works (figure 3). The audio signal is applied at phone-jack J1 to a non-inverting gain stage based on a 741-type operational amplifier. Potentiometer P1 adjusts input signal level, and is mounted remotely with J1. The gain is equal to one plus the ratio of R2 to R1, and with the values shown, the gain is 7.8. The back-to-back Zener diodes Z1, Z2 clip the signal to +5V peak-to-peak, which is the analog input range of the AD570JD.

I needed a 2 μ S positive-going pulse from some pin of slot #7. That was not available, but the DEVICE SELECT line at pin 41 of the peripheral connector "becomes active [low]... when the address bus is holding an address between \$C0n0 and \$C0nF, where n is the slot number plus \$8." This specification [Apple II Reference Manual, p. 109] determines the choice of locations \$C0F0 and \$C0F1 for initiating the conversion, and reading the answer, respectively. The 74LS00 quad NAND-gate is the address decoder, and either triggers (\$C0F0) the 555 one-shot (set for 2 μ S by R3, C4), or momentarily puts the

Figure 3: Apple II ADC Board



Listing 1

```

100 HOME
110 REM ---DEFINE UPPER LIMIT OF BASIC AREA;
120 REM ---LOWER LIMIT OF DATA BUFFER;
130 REM ---K IS A CONVERSION CONSTANT
140 REM ---FOR PLOTTING IN HI-RES.
150 HIEM: @191:BASE = 29552:W = 159 / 255
160 REM ---PROMPT.
170 PRINT "PLAY A FEW NOTES AFTER HEARING TONE"
180 REM ---BRIEF DELAY.
190 FOR I = 1 TO 500: NEXT
200 REM ---BRIEF TONE.
210 FOR I = 1 TO 200: E = PEEK (49200): NEXT
220 REM ---DEFINE JMP #0300 FOR USR FUNCTION.
230 POKE 10,76: POKE 11,0: POKE 12,3
240 REM ---CALL DATA COLLECTION ROUTINE.
250 ANS = USR (X)
260 REM ---INITIALIZE FOR SCREEN-PLOT ("PAGE") CAPTION.
270 TIME = 12: PAGE = 0
280 REM ---INITIALIZE FOR HI-RES:
290 SW = 49232
300 REM ---MIXED,
310 POKE SW + 3,0
320 REM ---PAGE-ONE,
330 POKE SW + 4,0
340 REM ---HIGH-RESOLUTION,
350 POKE SW + 7,0
360 REM ---GRAPHICS.
370 POKE SW,0
380 REM I---ADJUST CURSOR FOR CAPTION.
390 HOME : UTAB 21: POKE 36,0
400 REM ---DISPLAY CURRENT PAGE.
410 GOSUB 560
420 REM ---COMPUTE AND DISPLAY CAPTION.
430 LFT = PAGE * TIME
440 RIT = LFT + TIME
450 PRINT "TIME = "LFT" MS";
460 PRINT TAB( 14)"PAGE="PAGE;
470 PRINT TAB( 22)"TIME = "RIT"MS"
480 REM ---INTERPRET PRESSED KEY.
490 PRINT "PRESS E FOR EARLIER PAGE, L FOR LATER"
500 GET K$: IF K$ < > "E" AND K$ < > "L" THEN GOTO 500
510 IF K$ = "E" AND PAGE > 0 THEN PAGE = PAGE - 1: GOTO 390
520 IF K$ = "L" AND PAGE < 30 THEN PAGE = PAGE + 1: GOTO 390
530 GOTO 500
540 REM ---HI-RES SCREEN-PLOT;
550 REM ---INITIALIZE BEGIN AND END BYTES.
560 LO = BASE + PAGE * 280
570 HI = LO + 279
580 REM ---CLEAR PREVIOUS PLOT.
585 HGR
590 REM ---PLOT 280 DATA POINTS FROM BUFFER.
600 FOR I = LO TO HI
610 Y = PEEK (I)
620 Y = INT (Y * W)
630 HPLT I - LO,159 - Y
640 NEXT
650 RETURN

```

atched data at the converter output (SCOF1) onto the system data bus. That is how the circuit works.

While assembling the unit, I was concerned that the 12V would get through a wiring error onto some TTL line — either onto the +5V supply, or onto an address or data line. So I advise you to do at least as much double-checking and testing of the circuit as I did, even before plugging it in without chips. First, perform continuity tests to insure that there are no paths from -12V to +5V; second, see if all pins have "correct" resistance to ground. Third, leave all chips out, and install the board in slot #7. Turn on the Apple II and if it behaves normally, measure voltages at all pins of all sockets on the new unit. Fourth, when you are satisfied again that all is well, turn off the power, pull the board, and install U4, U5.

Fifth, put the board back, re-apply power, check voltages at the installed chips, and if all is well, try the following test program from the machine language monitor:

```

OC00 STA $C0F0
          STA $C0F1
          JMP $C000

```

Or, for more fun, try the slightly more elaborate

```

100 A = 49392: B = A + 1: X =
          40/255
110 POKE A,0
120 Y = PEEK(B): P = INT(X*Y)
130 PRINT TAB(P); "*"
140 GOTO 110

```

Be sure to try out the attenuator, P1.

Listing 2

```

;*****
;*
;* DIGITAL STORAGE *
;* OSCILLOSCOPE *
;*
;* ELLIS COOPER *
;*
;*****
;
SCOPE ORG #300
;
0300 48      START PHA      ;SAVE PROCESSOR
0301 8A      TXA          ; STATUS ON STACK...
0302 48      PHA
0303 98      TYA
0304 48      PHA
0305 08      RFP
0306
;
0306 A96F    LDA #86F      ;INITIALIZE TWO CONSECUTIVE
;          ;BYTES,
;          ;84C 84B, IN PAGE ZERO AS
;          ;POINTER TO DATA BUFFER..
0308 854B    STA 84B
030A A973    LDA #873
030C 854C    STA 84C
030E 8DFOC0  CNV1 STA $C0F0 ;START A CONVERSION.
0311 A207    LDX #807      ;DELAY UNTIL END
0313 CA      DELAY DEI     ; OF CONVERSION.
0314 D0FD    BNE DELAY
0316
;
0316 ADP1C0  LDA $C0F1      ;READ PULSE AMPLITUDE.
0319 C98C    CMP #88C      ;IS THRESHOLD EXCEEDED?
031B 30F1    BNE CNV1
031D
;
031D A000    LDY #800      ;YES; CLEAR Y FOR
;          ;INDIRECT-INDEXED MODE
031F 8DFOC0  CNV2 STA $C0F0 ;START A CONVERSION
0322 A54C    LDA 84C      ;LAST PAGE OF BUFFER?
0324 C994    CMP #894
0326 D006    BNE INCR     ;NO.
0328 A54B    LDA 84B      ;YES; LAST BYTE OF LAST PAGE OF
032A C900    CMP #800     ;OF BUFFER?
032C F015    BRQ EXIT     ;YES.
032E
;
032E 18      INCR CLC      ;NO; CLEAR CARRY FOR ADDITION
032F A54B    LDA 84B
0331 6901    ADC #801     ;INCREMENT THE POINTER...
0333 854B    STA 84B
0335 A54C    LDA 84C
0337 6900    ADC #800
0339 854C    STA 84C
033B
;
033B ADP1C0  LDA $C0F1      ;READ PULSE AMPLITUDE
033E 914B    STA ($4B),Y  ;STORE IN DATA BUFFER
0340 4C1F03  JMP CNV2     ;GO BACK FOR MORE
0342
;
0342 28      EXIT PLP      ;RESTORE PROCESSOR STATUS
0344 68      PLA          ;FROM STACK
0345 A8      TAY
0346 68      PLA
0347 AA      TAX
0348 68      PLA
0349
;
0349 60      RTS          ;RETURN TO BASIC.
;          ;END

```

If these little tests do not turn up any surprises, it is time to put in the main program of this article (listing 1 and listing 2). Be sure to save both parts before running. Be warned, you must have 48K RAM to use this software exactly as written. Like I said, though, you should use these listings only as a starting point, if at all, to carry out your own ideas.

One refinement of the software would be to display only every nth sample, or to sample less frequently but for a longer duration. Another idea is to swap back and forth between two high-resolution pages, achieving an "animated" display of the waveform. As for me, it is time to bone up on algorithms for extracting significant information from the stored data, e.g., pitch periods, envelopes, and so forth.

References

1. "The Piecewise-Linear Technique of Electronic Music Synthesis," E.D. Cooper and A.D. Bernstein, *Journal of the Audio Engineering Society*, V. 24, No. 6, July/August, 1976, pp. 444-454.
2. "Circuits for the Piecewise-Linear Technique of Electronic Music Synthesis," E.D. Cooper, *Electronotes Newsletter of the Musical Engineering Group*, V. 8, No. 69, September, 1976, pp. 8-13.
3. "Program Performs Harmonic Analysis," E.D. Cooper, *μComputerist Corner*, EDN, February 5, 1980, pp. 80-85.

Using an oscilloscope you should see 2 μs positive pulses at U2(11) and 0.5 μs active-low pulses at U3(1) and U3(19). Now, sixth, install the remaining chips and repeat all tests.

If everything appears OK at this point, you may be confident that your board is working, but there is nothing

like a conversion to convince you. Seventh, plug in an electric guitar or a microphone, or even just a speaker which has a big magnet, and try a program in BASIC:

```

100 POKE 49392,0
110 PRINT PEEK(49393)";
120 GOTO 100

```

Ellis Cooper owns an Apple II Plus microcomputer with a single disk drive, NEC 12" video monitor, and Centronics 737 printer. He is employed as a research mathematician by an international gold bullion dealer.