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MAY 1977/\$1.25

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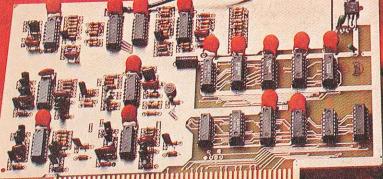
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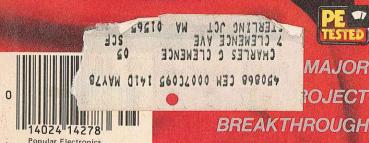
THE NEW "NO CAMERA" PC BOARD KITS

VOICE COMMUNICATIONS WITH



COMPUTERS!

...speech recognition device substitutes for keyboard



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MAJOR



Clockwise from top: CPU, 9" Video Monitor, Impact Printer, Keyboard, Cassette Storage System with Four Drives.

If you are seriously considering the purchase of a microcomputer system for personal or business use...or just beginning to feel the first twinges of interest in a fascinating hobby...the Digital Group is a company you should get acquainted with.

For many months now, we've been feverishly (and rather quietly) at work on our unique, high-quality product—a microcomputer system designed from the inside out to be the most comprehensive, easy-to-use and adaptable system you'll find anywhere. And our reputation has been getting around fast. In fact, you may have already heard a little something about us from a friend. We've found our own best salesmen are our many satisfied customers.

There's a good reason. Simply, the Digital Group has a lot to offer: state-of-the-art designs, a totally complete systems philosophy, unexcelled quality, reasonable software, affordable prices and the promise that our products will not become rapidly obsolete, even in this fast-moving, high-technology field.

The Advantages

Here are a few specific advantages of our product line:

- We offer interchangeable CPUs from different manufacturers (including the new "super chip" the Z-80 from Zilog) which are interchangeable at the CPU card level. That way, your system won't become instantly obsolete with each new design breakthrough. The major portion of your investment in memory and I/O is protected.
- Digital Group systems are complete and fully featured, so there's no need to purchase bits and pieces from different manufacturers. We have everything you need, but almost any other equipment can be easily supported, too, thanks to the universal nature of our systems.
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 With our combination of TV, keyboard, and cassette recorder, you have a system that is quick, quiet, and inexpensive. To get going merely power on, load cassette and go!
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 512 character upper & lower case video interface
 (1024 optional)
 100 character/second digitally synthesized audio
 cassette interface
- CPU Card
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 Vectored Interrupts (up to 128)
 256 byte 1702A bootstrap loader
 All buffering, CPU dependencies, and housekeeping circuitry
- Input/Output Card
 Four 8-bit parallel input ports
 Four 8-bit parallel output ports
- Four 8-bit parallel output ports

 Motherboard

Prices for standard systems including the above features start at \$475 for Z-80, \$425 for 8080 or 6800, \$375 for 6500.

More

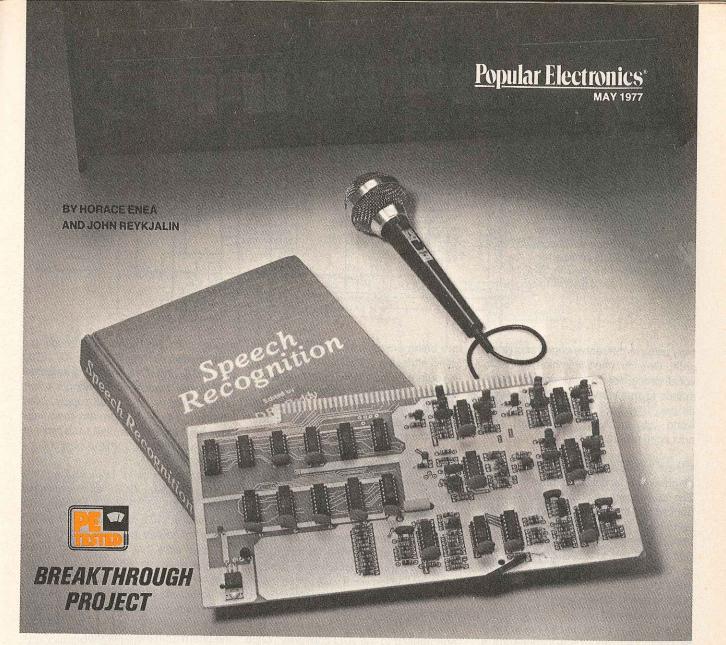
CIRCLE NO. 18 ON FREE INFORMATION CARD

Many options, peripherals, expansion capabilities and accessories are already available. They include rapid computer-controlled cassette drives for mass storage, printers, color graphics interfaces, memory, I/O, monitors, prom boards, multiple power supplies, prototyping cards and others. Software packages include BASICs, Assemblers, Disassemblers, Text Editors, games, ham radio applications, software training cassettes, system packages and more (even biorhythm).

Sounds neat - now what?

Now that you know a little about who we are and what we're doing, we need to know more about you. In order for us to get more information to you, please take a few seconds and fill in our mailing list coupon. We think you'll be pleased with what you get back.

POPULAR ELECTRONICS



INTRODUCING SPEECHLAB

THE FIRST HOBBYIST VOCAL INTERFACE FOR A COMPUTER!

Now your computer can respond to vocal commands by the simple addition of a \$250 single-board unit.

MAGINE being able to talk to your computer and have it respond by way of a hard-copy device or by activating some external appliance! Computer hobbyists can now enjoy this facility by building "Speechlab," a new, low-cost (under \$250) computer peripheral. To use it, all one does is plug the single Speechlab pc board into an Altair-bus connector (used by many microcomputer manufacturers), enter a special program, and the computer does the rest.

It's a state-of-the-art approach at a moderate cost.

One section of the program allows the user to "train" the computer to accept a vocal input (via a microphone), analyze the spoken word, and create a digitized version that is stored in memory. The second part of the program allows the user to speak to the Speechlab and have the computer generate the output selected for that particular sound.

The vocabulary size of Speechlab is a

function of the speech recognition algorithm used and the amount of memory available. For the program used in this article, it is 64 bytes per spoken word.

The unique characteristics of Speechlab open many formerly closed doors. Since Speechlab will operate with any audio input (not necessarily a recognized language), a person who's vocally handicapped can operate almost any number of appliances (TV receiver, stereo system, solenoid-operated door,

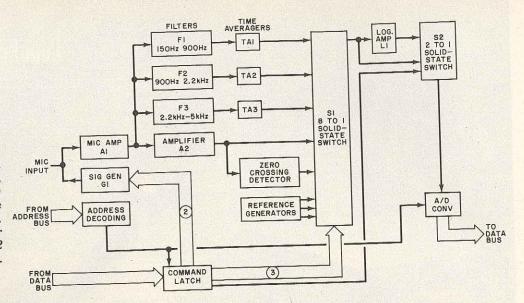


Fig. 1. The mic input is amplified, filtered and applied to S1 along with raw audio, zero-crossing detection, and three detection, and three reference voltages. ADDRESS BUS Output of S1 is computer selected by switch S2 for digitizing.

etc.) using a repeatable sound such as a grunt. One can use Speechlab, too, as a vocal processor to add spoken commands to many computer games (such as the "Star Trek" game), or enter the world of artificial intelligence and advanced programming.

Circuit Operation. The basic block diagram of Speechlab is shown in Fig. 1. The audio input is amplified by A1 and applied to three 80-db/decade rolloff band-pass filters F1, F2, and F3. These filters encompass the ranges of 150 to 900 Hz, 900 Hz to 2.2 kHz, and 2.2 kHz to 5 kHz, respectively. These ranges correspond to the frequency ranges of the first three resonances of the average human vocal tract.

Each filter is passed to a time averager (TA1, TA2, and TA3) to generate a voltage proportional to the level of the speech waveform within each band.

The amplified audio signal from A1 is further amplified by A2 to generate an unfiltered waveform that can swing ±2 volts about a rest level of 2 volts. This signal is also applied to a zero-crossing detector that generates a voltage proportional to the number of times the speech waveform crosses the 2-volt rest level in a given period of time, thus generating a measure of the dominant frequency in the speech signal.

These five voltages—TA1, TA2, TA3, A2, and ZCD-are fed to solid-state switch S1 along with three reference voltages used for calibration and self test. A computer output command selects one of these five voltages to be passed through S1.

The selected output from S1 is passed to a second solid-state switch (S2), and to a logarithmic amplifier (L1) that emphasizes the low-level signal be-

fore being passed to S2. Switch S2 can select either the direct output from S1, or the output from L1, and pass this selected signal to a 6-bit A/D converter where the voltage is converted to a digital value. The output of the A/D converter is fed to the computer data bus.

All operations of the Speechlab are controlled through a single I/O port (address AFhex). As shown in Fig. 2, six bits are used: bit-5 disables the 8-to-1 multiplexer (S1), and is used when switching between bands; bit-4 controls signal generator G1 which is used either to drive the microphone so that it acts like a miniature loudspeaker for prompting during voice input, or to drive the filters and zero-crossing detector during calibration and test; bit-3 selects either linear or logarithmic scaling of the voltage applied to the A/D converter; while bit-2, bit-1, and bit-0 select one of the eight signals from S1 for A/D conver-

The input data word contains the 6-bit A/D output in bits 0 through 5, bit-6 is unused and is always 0, while bit-7 is the A/D converter status with a 1 corresponding to busy, and 0 corresponding

Speechlab is physically configured to occupy one slot in the Altair bus, and the complete schematic is shown in Fig. 3 through Fig. 7.

Construction. The two foil patterns (Speechlab uses one double-sided pc board) are shown half-size in Fig. 8. (Blow up to full size on film only.) Component layout is shown in Fig. 9.

All the components are mounted on one side of the board, with all the soldering done on the noncomponent side. Sockets are recommended for all IC's since most of them are MOS-types that

may be damaged by improper handling.

Integrated circuits IC1, IC4, IC7, IC8, IC9, IC15, and IC16 should be selected so they are capable of delivering a 4-volt output when using a 5-volt supply. Dual flip-flop IC14 can be from any manufacturer but Fairchild, as their truth table is somewhat different from the convention-

Start construction by installing the voltage regulator (IC6), all the discrete components, and the IC sockets-do not install the IC's at this time. Check the board for correct parts installation, and to make sure that there are no solder bridges between adjacent foil traces. Mount the board in an Altair bus connector, and check for the presence of 5 volts at the output of the voltage regulator and at the appropriate socket pins. Remove the board from the computer.

Install IC2 through IC5, IC10 through IC14, and IC17 through IC22. Install the board back in the Altair bus connector, and turn on the computer. Load the test

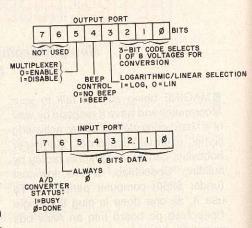
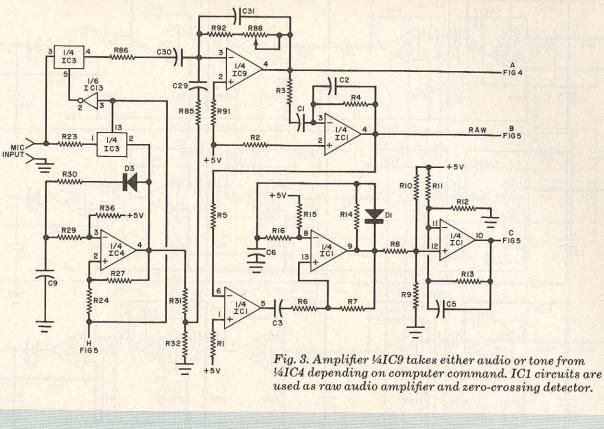


Fig. 2. Input and output port bit configuration.



	PARTS LIST
Unless otherwise noted, the following capaci-	L1—22-μH choke
tors are 10% Mylar types, and all picofarad	Unless otherwise noted, the following resistor
sizes are CM05 types.	are 1/4-W, 5%
C1, C16, C21, C43, C47, C49, C52,	R1—619,000 ohms, 1%
C57—0.0047 µF	R2—1 megohm, 1%
C2, C31—100 pF	R3—6810 ohms, 1%
C3, C17, C20—270 pF	R4—332,000 ohms, 1%
C4, C7, C8, C10, C12, C19, C27, C32, C33,	R5—200,000 ohms, 1%
C34, C35, C36, C37, C44, C55, C61,	R6,R20,R21—30,000 ohms
C62—0.1 µF, 25-V disc	R7, R100—3 megohms
C5, C14, C18, C24, C54, C60—0.01 µF	R8, R9, R10, R12, R14, R16, R104—1 me
C6, C42, C45, C53, C56—240 pF	gohm
C9, C40, C48—0.022 µF	R11—910,000 ohms
C11, C29—47 pF	R13—2.7 megohms
C13—15 µF, 25 V tantalum	R15, R48—10 megohms
C15, C22, C51, C59—0.0015 µF	R17,R18—20,000 ohms
C23—0.0022 µF	R19, R22, R106—10,000 ohms
C25, C26, C28, C38—1 µF	R23—1000 ohms
C30, C39, C46—0.047 µF	R24, R27—1.2 megohms
C41—0.1 µF	R25, R34, R39—470,000 ohms
C50, C58—0.001 µF	R26, R38—750,000 ohms
D1, D3 through D6—1N4148 or 1N914 diode	R28, R31—100,000 ohms
D2—1N746 diode	R29—110,000 ohms
IC1, IC4, IC7, IC8, IC9, IC15, IC16—	R30—39,000 ohms
LM3900 quad amp	R32-47,000 ohms
IC2—4051 8-to-1 analog multiplex	R33, R41—68,100 ohms, 1%
IC3—4016 quad analog switch	R35, R96,R102—75,000 ohms
IC5—LM311 comparator	R36—3.9 megohms
IC6—78M05 5-volt regulator	R37, R46—357,000 ohms, 1%
IC10—4024 7-stage binary counter	R40, R50, R52, R54, R56, R58, R60
IC11, IC18—74C174 D flip-flop	R61—10,000 ohms, 1%
IC12—4050 hex buffer	R42—12,100 ohms, 1%
IC13, IC22—4049 hex buffer inverter	R43, R49—4750 ohms, 1%
IC14—4013 (see text) dual-D flip-flop	R44 4320 ohms, 1%
IC17—74LS30 8-input NAND gate	R45, R47—681,000 ohms, 1%
IC19—8097 three-state hex buffer	R51, R53, R55, R57, R59-4990 ohms, 1%
IC20—8093 three-state quad buffer	R62—274,000 ohms, 1%
IC21—4001 NOR gate	R63—7500 ohms
MIC-Mura DX-121 dynamic microphone	R64, R66, R72, R75—160,000 ohms
(part of stereo set Mura DX-242)	R65, R71—12,000 ohms

Anioem	
22-µH choke	R67, R70—300,000 ohms
ss otherwise noted, the following resistors	R68—931,000 ohms, 1%
1/4-W, 5%	R69—2 megohms
619,000 ohms, 1%	R73—620,000 ohms
1 megohm, 1%	R74, R76, R90, R92—62,000 ohms
6810 ohms, 1%	R77—15,000 ohms
332,000 ohms, 1%	R78, R83, R84—147,000 ohms, 1%
200,000 ohms, 1%	R79, R80, R87—51,100 ohms, 1%
20,R21—30,000 ohms	R81,R82,R89—174,000 ohms, 1%
R100—3 megohms	R85—330,000 ohms
R9, R10, R12, R14, R16, R104—1 me-	R86—680 ohms
hm	R88—100,000-ohm pc trimmer potentiomete
–910,000 ohms	R91—270,000 ohms
–2.7 megohms	R93—249,000 ohms, 1%
R48—10 megohms	R94—4300 ohms
R18—20,000 ohms	R95, R97, R103, R105—360,000 ohms
R22, R106—10,000 ohms	R98, R101—820,000 ohms
-1000 ohms	R99—845,000 ohms, 1%
R27—1.2 megohms	R107—158,000 ohms, 1%
R34, R39—470,000 ohms	R108—4700 ohms
R38—750,000 ohms	R109, R111, R117, R119—82,000 ohms
R31—100,000 ohms	R110, R116—5100 ohms
–110,000 ohms	R112, R115—180,000 ohms
–39,000 ohms	R113—549,000 ohms, 1%
-47,000 ohms	R114—1.6 megohms
R41—68,100 ohms, 1%	R118—510,000 ohms
R96,R102—75,000 ohms	R120—6800 ohms
-3.9 megohms	R121—2000 ohms
R46—357,000 ohms, 1%	Misc.—Sockets (one 8-pin, thirteen 14-pin
R50, R52, R54, R56, R58, R60	seven 16-pin), regulator mounting hard
1—10,000 ohms, 1%	ware, tie-wrap etc.
-12,100 ohms, 1%	Note 1: The following is available from Heu-
R49—4750 ohms, 1%	ristics Inc., 900 N. San Antonio Rd. (Suite
-4320 ohms, 1%	C-1), Los Altos CA 94022 (Tele
R47—681,000 ohms, 1%	415-948-2542): complete kit of all parts in-
R53, R55, R57, R59—4990 ohms, 1%	cluding pc board, sockets, microphone
-274,000 ohms, 1%	hardware manual, and 200-page lab manu-
-7500 ohms	al, SpeechBasic, and assembly language
R66, R72, R75—160,000 ohms	programs at \$249. (California residents
R71—12,000 ohms	please add 61/2% sales tax.)

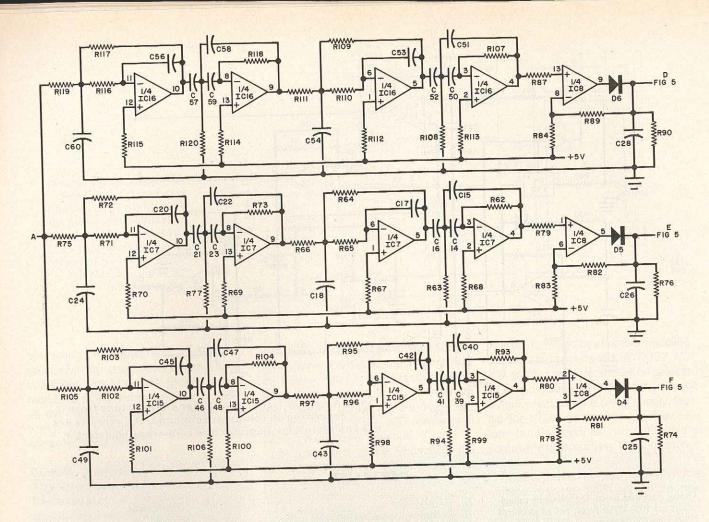


Fig. 4. Three bandpass filters and their associated time averagers. They encompass three ranges corresponding to frequency ranges of the first three resonances of an average human vocal tract.

program of Table I at 100 (hex). NOTE: all program data in this article is in hex.

You must jump to your monitor routine at address 0164-0165. Load address 195 with 05 and run the program. This will input the fixed reference voltage levels to the A/D converter and check the signal paths from switch S1 to the computer data bus.

After running this program, examine locations 200 through 20F, 300 through 30F, and 400 through 40F. Location 200 through 20F should contain 12 ± 4 , 300 through 30F should contain 24 ± 4 , and 400 through 40F should contain 36 ± 4 .

Insert the remaining IC's in their sockets, load location 195 with 10, and run the test program (Table I). This test uses the signal generator (G1) to create an input for the filters, amplifiers, and zerocrossing detector, and thereby checks the remaining signal paths on the board and calibrates the microphone preamplifier. After running the program, examine locations 200 to 20F to see if it contains 16 to 18. If not, adjust potentiometer R88 and rerun the program until these outputs occur.

Calibration and Test Program.

The test program (Table I) is a generalpurpose calibration, test, and diagnostic program for the Speechlab. It loads at location 100 and requires memory from 100 to 600 for program and data areas. Locations 163-165 should be loaded with a jump to your monitor address so that the program will return control to your monitor after execution. If you do not have a monitor, place a halt at this location.

The program collects four 256-byte buffers of data from four of the eight pos-

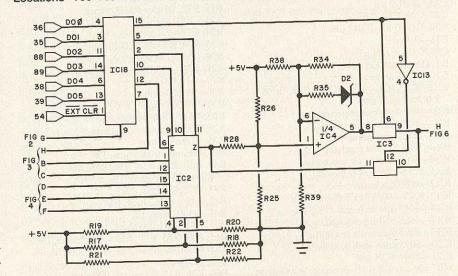
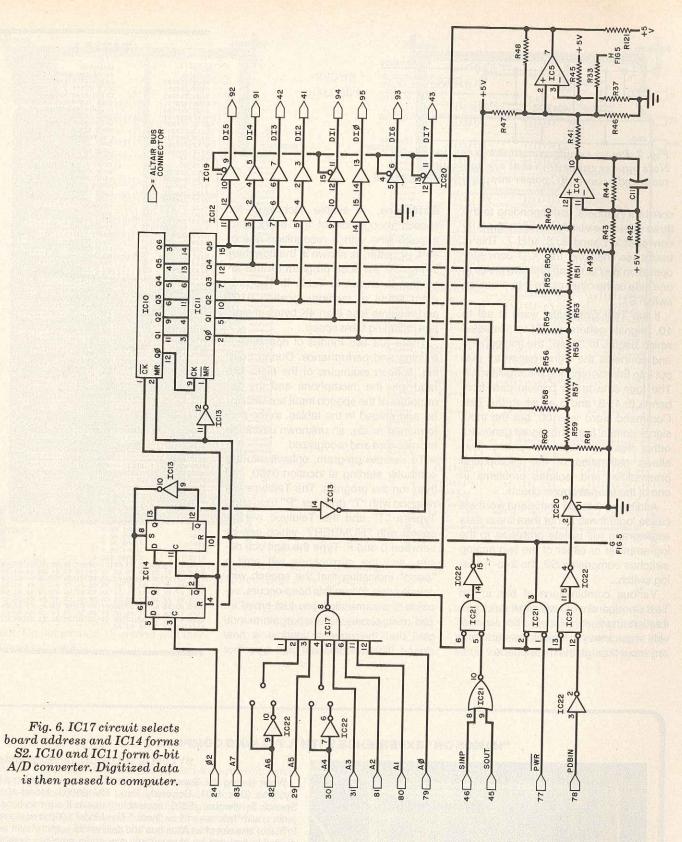


Fig. 5. Command latch (IC18) can activate tone generator and switch S1 (IC2). Op amp (¼IC4) is logarithmic amplifier.



sible inputs to the A/D converter. The first of the four bands is specified by the Test Command word, which also specifies beeper on/off and linear or logarithmic scaling. The next three bands are 1, 2, and 3 greater than specified by the Test Command word. Each band is sampled every five milliseconds until 256 samples have been collected from

each of the four bands. Data from the first band is stored in 200 to 2FF, the second band from 300 to 3FF, the third from 400 to 4FF, and the fourth from 500 to 5FF.

For example, if the Test Command word is set to 00, after the test program is run, the four data areas will contain numbers representing the outputs of band-0 (low frequency), band-1 (mid frequency), band-2 (high frequency), and band-3 (zero-crossing detector). Anything that was spoken into the microphone while the program was running, is filtered, converted into a binary number, and stored in the data areas.

If the Test Command word is set to 05, the first three data areas will contain

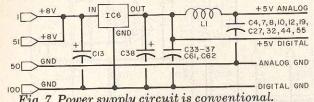


Fig. 7. Power supply circuit is conventional.

Note bypass capacitors that are actually mounted between IC power supply pins and ground.

constant numbers corresponding to the three reference voltage levels to the A/D converter on band 5, 6, and 7. This is useful for checking the A/D converter operation and isolating problem areas to one side or the other of the 8-to-1 analog switch S1.

If the Test Command word is set to 10, signal generator *G1* is enabled which begins to "beep" the microphone and connects the signal-generator output into the microphone preamplifier *A1*. The four data areas contain data from bands 0, 1, 2, and 3 as when the Test Command word was 00, but the input signal comes from the signal generator rather than from the microphone. This allows calibration of the microphone preamplifier and isolates problems in one of the filter-averager chains.

Adding bit-3 to the command word will cause logarithmic rather than linear data scaling and will isolate problems to the log amplifier or either of the two analog switches comprising S2, the 2-to-1 analog switch.

Various combinations of bits in the Test Command word will allow quick calibration and fault isolation, and also provide a quick way to look at raw data from any input through the microphone.

Software. A simple technique for speech recognition of the digits zero through nine with a recognition rate of 90% or better, is shown in the flowchart of Fig. 10. An 8080 program for this algorithm is shown in Table II. The program starts at memory location 0100 and requires less than 4K bytes of storage including table space.

There are two modes of operation—training and performance. During training, speech examples of the digits are read into the microphone and the parameters of the speech input are extracted and placed in the tables. In the performance mode, an unknown utterance is presented and recognized.

To use the program, enter it into the computer starting at location 0100, and then run the program. The Teletype will respond with "T" (train) or "P" (perform). Type a "T" and the Teletype will respond with "NUMBER?" which can be between 0 and F. Type the digit you desire, and the microphone will emit a "beep" indicating that the speech window is open. When this beep occurs, vocalize the same digit you just typed in. The microphone will beep again to indicate that the speech window is now closed. The machine will then type T or

TABLE

0100					RG										
0100	=			S	TAR	I E	TAR	T+1	OOF						
	210002				HLD				001	0					
0103	228D01 210003			ĭ	XI	H,S	TAR	T+2	100	1					
0109	228F01				HLD										
	210004			L	XI	H,9	TAR	T+3	1001	1					
010F	229101			5	HLD	TE	HP3	5							
	210005			L	IX.	H, 9	TAR	T+4	1001	1					
	229301			5	HLD	TE	HP4								
0118	3A9501		GO	L	DA	COM	ANI)							
	CD6601			C	ALL	IV	PUT								
	2A8D01			1	HLD	TE	MP1								
0121	77			,	VOI	MAR									
0122	2C			1	NR SHLD		ND4								
	228D01				DA	CUP	DEL								
0126	3A9501			-	DI	1	HINT								
0129	CD6601			- 7	ALL	TA	PHI	-							
	2A8F01			ì	HLD	TE	MP	,							
0131	77														
0131					INR SHLI	L									
0132	2C 228F01 3A9501				SHLD	TE	MP2	2							
0136	349501			1	DA	CON	IANI	0			100				
0130	C602			-	ADI	2									
	CD6601			-	CALL	II.	(PU	r							
	2A9101			1	HLI	TE	MP.	3							
0141				- 1	Unw	24 - 4	3								
0142	20				INR	L									
0143	2C 229101				SHLI	TI	MP:	3							
0146	3A9501			- 1	LDA	COL	IAN	D							
0149	C603				ADI	3		300							
	CD6601			- 1	CALL	. II	APU.	T							
014E	2A9301				LHLI	TI	MP	4							
0151	77 20 229301				VOM	Hri	4								
0152	2C				INR	L									
0153	229301	MIL			SHLI			4							
0156	CA5F01	170			JZ S	STO									
				4.5	CALI	- 10	LA	1							
	C31801		120000		JMP	GU									
015F	3E00		STOP		UUT	OA									
0161	D3AF C3XXXX	UD.			JMP			м							
0163	C3XXXX		******												
0166	F620		INPUT		ORI	0A	EH								
	D3AF				THA	On	-0								
016A	E6DF				ANI	04	FH								
0166	D3AF DBAF		LOOP		IN	DAF	н								
0170	17		Looi		RAL										
0170	DAGEO:	1			JC I	nn	P								
0174	DBAF	18.			IN	OAF	Н								
0176					RET										
0177			DELAY		PUS	н в									
0178	3E05				MVI	Ar	5								
017A	FEOO		DELO		CPI	0									
0170		1			JZ	RET	DEL								
017F	0669	3			HVI	Br	69H								
0181			DEL1		NOP										
0182	00				NOP										
0183	00 05 C2810				DCR	B									
0184	C2810	1			JNZ		L1								
0187	3D C37A0				DCR										
		1			JMP		LO								
0181	C1		RETDE	EL	POP	B									
0180	C9				RET										
C000) =		SYSTE			000	100	1							
0181			TEMP:	I DE	2										
018			TEHP:	2 DS	3 2										
0191			TEMP:												
0193			TEMP			-									
	YY		COMA	AD E	au	\$	^-	22	or	04	24	00	04	22	
	21 00		22 BD				03		8F	01	21				
0110	91 01	21		22	93	01	3A	95 C6	01	CD	66	01	2A 2A	8D	
0120		20	22 BD	01	3A	95 95	01	C6	01	CD	66	01	2A		
0130		20	22 8F 22 91	01	3A	95	01	C6	03	CD	66	01	2A	93	
0140			22 91		CA	5F	01	CD	77	01	C3	18	01	3E	
0150			C3 53			20	D3	AF	77 E6	DF	D3	AF	DB	AF	
017			01 DR	AF	C9	C5	3E	05	FE	00	CA C9	88	01	06	
018	69 00	00	01 DB 05 C2	81	C9 01	3D	C3	7A	01	C1	C9	9B	F7	FO	
	FE 57		49 E2	A6	7	ATTEN OF	TORK,								

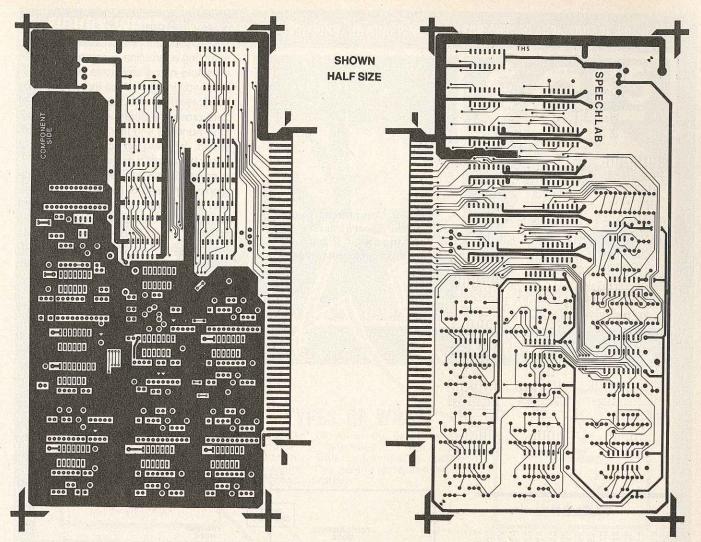


Fig. 8. Etching and drilling guides for pc board are shown half size. Guide at left is the component side. Component layout is in Fig. 9.

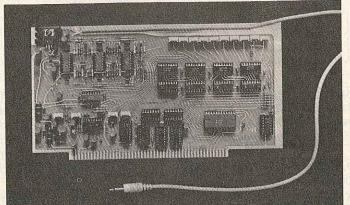
P again. You answer with a T, and the process is continued as long as you want. Do not exceed 16 entries with this sample program.

Once you have some vocalized digits in memory, run the program again. This time, when the Teletype asks T or P, answer with a P (for perform). Now, as you

speak the digits into the microphone, the Teletype will respond by typing that digit. When used in a quiet room, with the same vocalization, this algorithm can be

"HANDS ON" EXPERIENCE WITH A TALKING COMPUTER

BY LESLIE SOLOMON, Technical Editor



While testing the Speechlab, we borrowed an Al Cybernetic Systems (Box 4691, University Park, NM 88003) Model-1000 Speech Synthesizer (\$325, assembled) to see if our microcomputer could "talk" as well as "hear." The Model 1000 is designed to fit into one slot of an Altair bus and delivers its output via an audio cable that can be plugged into any audio amplifier system. The output level is 0.6 volt p-p; impedance is 1000 ohms; and frequency range is 150 to 4500 Hz.

This synthesizer is phoneme-oriented. Accordingly, you can program it to say anything, as opposed to speech synthesizers that have only several words fixed in ROM. Essentially, the Model 1000 is a hardwired analog of the human vocal tract and various portions of the circuit emulate the vocal cords, the lungs, and the variable-frequency resonant acoustic cavity of the mouth, tongue, lips and teeth.

All the information necessary to perform the synthesis functions are located within a ROM that is accessed by the program. Words and sentences are formed by supplying a string of ASCII characters as would be done when outputting to any port, except that these strings also use some non-alphanumeric characters (i.e., the "+" is used to form "th" as in "thaw" or "earth"). Each ASCII character represents a particular phonetic sound or phoneme. If desired, you can create a program that produces simultaneous printout and "voiceout" of the same string.

The device requires very little software to implement: less than 50 bytes of assembly language or a handful of BASIC statements. The manual accompanying the synthesizer covers speech generation in detail, how it is created, and what is involved. It also illustrates how to "mechanize" speech, with several examples shown.

After working with the synthesizer for a couple of weeks, we

found that we have a lot to learn about how humans create speech. After many hours of studying, experimenting, and redoing programs, we made the Model-1000 utter some recognizable sentences. It is not easy, our experience showed, even when one uses the wealth of instructions provided.

Working with a phoneme-oriented speech synthesizer is a little like learning to use a microprocessor. All the logic is there, but programming it properly is another story. Like working with a processor for the first time, one must crawl frustratingly before walking. Slowly, however, the ideas start to percolate. Our computer still talks with a rather heavy "robotic" accent, but we have hopes that someday it will "humanize."

To paraphrase Sam Johnson: "Sir, a computer talking is like a dog walking on its hind legs. It is not done well; but you are surprised to find it done at all." We have a long road ahead to the "HAL-9000," but the first step has been taken.

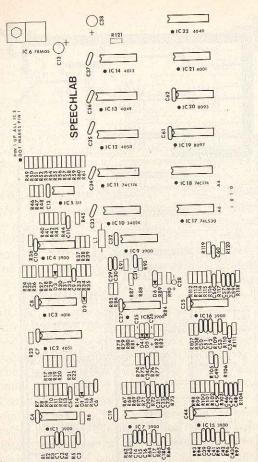


Fig. 9. Component layout for the Speechlab. See etching and drilling guide on previous page.

TABLEII

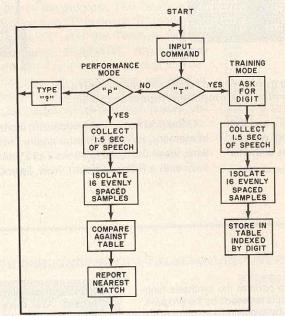


Fig. 10. Flow chart of a simple program that is used to "T" (train) and "P" (perform) a vocal operation. The program is shown in Table II.

plate with the smallest difference from the sample is then selected as the answer (output).

expected to have a recognition rate

The program works as follows: the sampling subroutine is entered to obtain a sample of the amplitude every 10 mil-

liseconds in each of the three frequency bands and to estimate the number of

zero crossings during each time period. One hundred and fifty samples are collected, allowing up to 1.5 seconds of

speech (between microphone "beeps"). A preset threshold is used to find the be-

ginning and end of the word. The dura-

tion of the word can now be computed

by a simple subtraction. Typically, this

duration will be about 400-milliseconds

for the digits. The duration time is divid-

ed by 16 to select 16 evenly spaced pa-

rameters from the three bands and zero-

The 64 bytes obtained (16 parameters from each of the four bands) are com-

pared with similar parameters which

were collected during the training mode. A summation (running total) of the differ-

ence between the 64 parameters of the

sample and the parameters of the training "templates" is computed. The totals

represent a measure of the difference

between the sample and each of the

previously stored templates. The tem-

crossing information.

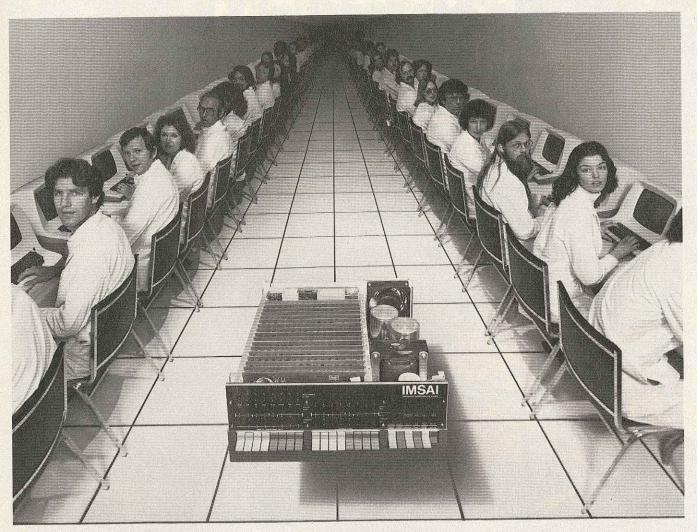
greater than 90%.

The above algorithm, while relatively simple, illustrates many of the basic concepts of speech recognition. A manual supplied with the Speechlab kit contains descriptions of other approaches to speech recognition, along with sample programs to demonstrate the techniques of speech recognition.



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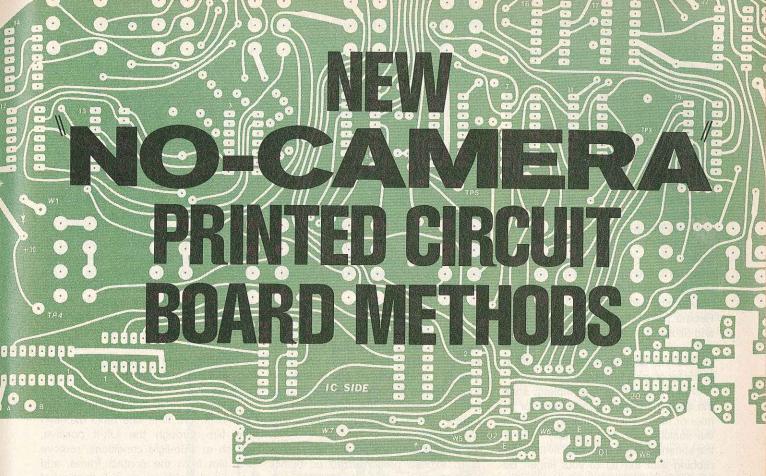
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