

CT-1
Hardware User's Manual

Computaker Consultants
P.O. Box 1951
Santa Monica, California 90406



SYNTHESIZING SPEECH BY RULE WITH THE COMPUTALKER MODEL CT-1

Synthesis-by-Rule is a method of producing synthetic speech which is considerably easier than computer/hand analysis of recorded human speech. The word or phrase to be synthesized is entered in the form of a phonetic code to a software system which generates the control parameters for the CT-1 Synthesizer board. The result is speech which is understandable to most people in all but the most difficult perceptual situations with high noise levels or speech material having completely unexpected content.

The demonstration cassette contains a portion of the Gettysburg Address synthesized using a system of software rules. Such a set of software acoustic-phonetic rules is available from Computalker Consultants coded for the 8080 CPU. This software system accepts a string of ASCII coded phonetic symbols with stresses marked, and produces a set of control parameters for the Model CT-1 Synthesizer. The example on the cassette was generated using a previous version of this software system, coded in FORTRAN, and running on a DEC PDP-12. As the parameter data was generated, it was punched on paper tape in the data format as described in the CT-1 Hardware User's Manual, and then read into the IMSAI 8080 for playback. That program, as run on the larger machine, was originally written for a different speech synthesizer, and some parameters required special treatment for conversion to the CT-1 parameter format. In some cases, this conversion was not accurately fine-tuned for the CT-1, and the direct output of the 8080 version of the program is somewhat clearer in some of the fine details.

The CSR1 Synthesis-by-Rule software system is organized around the philosophy of attempting to produce natural sounding, human quality speech, rather than trying to produce a stereotypical robot-like sound. Because the true structure of real human speech is not yet correctly represented in the software rules, the resulting speech sometimes has an eerie quality that makes the listener try to assign human-like traits and qualities to the "speaker" behind the voice. This psychological reaction to the voice does not occur when it is synthesized in a "robot" stereotype having little or no pitch variation and abrupt, blocky formant frequency transitions. The pitch control parameter (F_0) can easily be held to a constant value if the speech output sounds better to you that way. The CSR1 software system is structured around phonological, phonetic and acoustic principles in such a way that it can be modified to keep pace with the state of the art of synthesis of natural speech. The Model CT-1 has been designed as a general acoustic synthesizer so that the hardware will not pose limitations to further improvements in the obtainable speech output quality.

The CSR1 software system is set up as a general callable subroutine which accepts a string argument containing the phonetic text, and on completion, plays the speech data in the buffer directly to the CT-1. With this structure, CSR1 may be called either from a keyboard input loop (supplied with the code) giving an on-line phonetic synthesizer, or from another system such as BASIC or an operating system, which passes a stored or computed string argument containing the material to be synthesized. On return, the buffer contains the actual CT-1 data as synthesized, which may be written out to cassette or paper tape for editing with the CTMON Monitor/Editor program. The 8080 assembly code version of CSR1 fits in less than 6K bytes of memory, including all phoneme feature and target tables. This code may be located in ROM or RAM. Additional RAM will be required for parameter data storage during the actual synthesis. The buffer space required is 900 bytes per second of speech. By comparison, the introductory phrase, "Hello, I'm Computalker, A speech synthesizer designed to plug into the standard bus on your 8080 microcomputer" is less than 7K bytes long. CSR1 version 1.0 completes the computation of parameter data before beginning playback. An interrupt driven version is currently under development, which will begin playback as soon as sufficient data has been computed and stored in the buffer.

HOW TO GET NATURAL SOUNDING SPEECH OUTPUT FROM THE COMPUTALKER MODEL CT-1

The demonstration cassette, "Sounds of Computalker", illustrates several methods of obtaining the control parameters to operate the Computalker Model CT-1 Speech Synthesizer. High quality speech output, as exemplified by the introductory phrases, "Hello, I'm Computalker. A speech synthesizer ...", involves computer processing of recorded human speech followed by a fair amount of hand work. The recordings were initially digitized at 10K samples/second and then analysed using a linear prediction algorithm to extract the formant frequencies, and a cepstrum algorithm to measure the fundamental frequency. These techniques are described in several texts on speech analysis (Flanagan, J.L., Speech Analysis, Synthesis, and Perception, 2nd Ed., Springer Verlag 1972; Markel, J.D. and Gray, A.H., Jr., Linear Prediction of Speech, Springer Verlag 1976). In addition to these analyses, the amplitude was measured by RMS averaging a smooth window each 10 msec. to obtain the AV parameter. Some editing of the formant frequency data was done by hand to eliminate falsely detected peaks and fill in occasional gaps in the true formant data before converting the frequency data to the Computalker parameters F1, F2, and F3. Since the CT-1 control parameters consist of numerical values within the range of 0-255, all frequency and amplitude data is converted so that it stays within this range. All the above steps required approximately 6 hours of time on a DEC PDP-12 set up for speech analysis processing to produce the original data for the introductory phrases on the cassette. At this stage, this data was punched on paper tape and then read into the CT-1 Control Monitor program running on my IMSAI 8080. From that point, I spent several more evenings entering the data for parameters AH, AF, FF, and AN, and a bit more touching up of the other parameters.

Given the frequency vs. time information obtained from the initial computer analysis, the remaining aspiration and frication data can be inserted by fairly straight-forward procedures. These procedures will be described in the completed CT-1 Hardware User's Manual. The Manual will also discuss the approximate formant frequency patterns needed to construct the sounds of the various phonemes of English. It would be feasible (although tedious work) to construct intelligible sounds by hand editing based on this data. However, it is still quite difficult to form these patterns to make natural sounding speech without access to a spectrum analysis process of some kind. Such an analysis gives you the frequency structure as a function of time, i.e. retaining the natural timing structure.

It is my plan to publish more extensive descriptions of the above mentioned speech analysis techniques, to make them accessible to a wider audience than they now have. The recent developments in floating point hardware with multiplication in the 50-100 microsec. range make it reasonable to do this sort of analysis on a microcomputer. The setup would require a filter and A/D converter capable of sampling the speech at at least 10K samples/sec. The low-pass speech filter ahead of the A/D converter should be reasonably flat to at least $\frac{1}{3}$ of the sampling rate, and then down by at least 30-40 db at $\frac{1}{2}$ the sampling rate. 32K of RAM memory would allow sampling up to 3 seconds continuously, which is a workable sized chunk. Without floating point hardware, the analysis would proceed quite slowly, but in many cases, that is not a drawback on a micro system.

Alternatively, for a modest consulting fee, Computalker Consultants could supply the basic, rough formant frequency, F0 and AV data from your tape recording, leaving out the aspiration, frication and nasal values, which must be added by hand. As a preliminary estimate, I believe this work could be done for approx. \$25 per second of speech material to be analyzed. Working from this basic data, the desired speech could be produced following the tables and information given in the CT-1 Hardware User's Manual, using the CTMON Monitor/Editor to synthesize speech from the data as the work progresses.

Friends, Humans, and Countryrobots: Lend me your Ears

D Lloyd Rice
Computalker Consultants
821 Pacific St #4
Santa Monica CA 90405

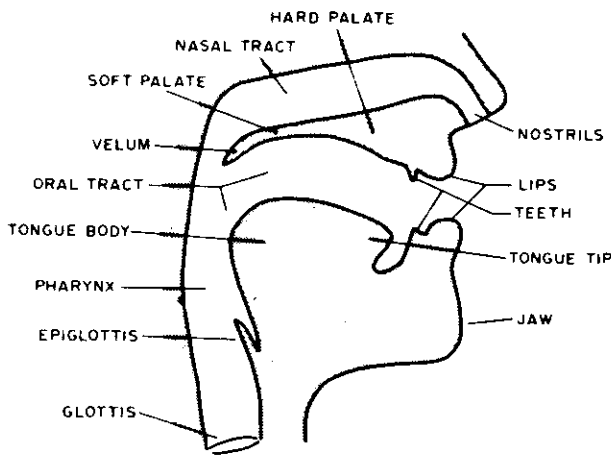


Figure 1: The Human Vocal Tract. The human vocal tract is roughly described as a tube approximately 17.4 cm long with varying resonance characteristics as muscles control the shape. The tract splits into two parts, nasal and oral, at the top, with a valve called the velum providing flexible control of the nasal resonances in given utterance. An electronic model of this natural organ roughly parallels the function of the tract.

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You've got your microcomputer running and you invite your friends in to show off the new toy. You ask Charlie to sit down and type in his name. When he does, a loudspeaker on the shelf booms out a hearty "Hello, Charlie!" Charlie then starts a game of Star Trek and as he warps around thru the galaxy searching for invaders, each alarming new development is announced by the ship's computer in a warning voice, "Shield power low!", "Torpedo damage on lower decks!"

The device that makes this possible is a peripheral with truly unlimited applications, the speech synthesizer. This article describes what a speech synthesizer is like, how it works and a general outline of how to control it with a microcomputer. We will look at the structure of human speech and see how that structure can be generated by a computer controlled device.

How can you generate speech sounds artificially, under computer control? Let's look at some of the alternatives. Simplest of all, with a fast enough digital to analog converter (DAC) you can generate any sound you like. A 7 or 8 bit DAC can produce good quality sound, while somewhere around 4 or 5 bits the quantization noise starts to be bothersome. This noise is produced because with a 5 bit data value it is possible to represent only 32 discrete steps or voltage levels at the converted analog output. Instead of a smoothly rising voltage slope, you would get a series of steps as in figure 2. As for the speed of the DAC, a conversion rate of 8,000 to 10,000 conversions per second [The sample rate in conversions per second or samples per second is often quoted in units of Hertz. We will use that terminology here, although conversions

per second is a generalization of the concept of cycles per second] is sufficient for fairly good quality speech. With sample rates below about 6 kHz the speech quality begins to deteriorate badly because of inadequate frequency response.

Almost any microprocessor can easily handle the data rates described above to keep the DAC going. The next question is, where do the samples come from? One way to get them would be by sampling a real speech signal with a matching analog to digital converter (ADC) running at the same sample rate. You then have a complicated and expensive, but very flexible, recording system. Each second of speech requires 8 K to 10 K bytes of storage. If you want only a few words or short phrases, you could store the samples on a ROM or two and dump them sequentially to the DAC. Such a system appears in figure 3.

If you want more than a second or two of speech output, however, the amount of ROM storage required quickly becomes impractical. What can be done to minimize storage? Many words appear to have parts that could be recombined in different ways to make other words. Could a lot of memory be saved this way? A given vowel sound normally consists of several repetitions of nearly identical waveform segments with the period of repetition corresponding to the speech fundamental frequency or pitch. Figure 4 shows such a waveform. Within limits, an acceptable sound is produced if we store only one such cycle and construct the vowel sound by repeating this waveform cycle for the duration of the desired vowel. Of course, the pitch will be precisely constant over that entire interval. This will sound rather unnatural, especially for longer vowel durations, because the period of repetition in a naturally spoken vowel is never precisely constant, but fluctuates slightly. In natural speech the pitch is nearly always changing, whether drifting slowly or

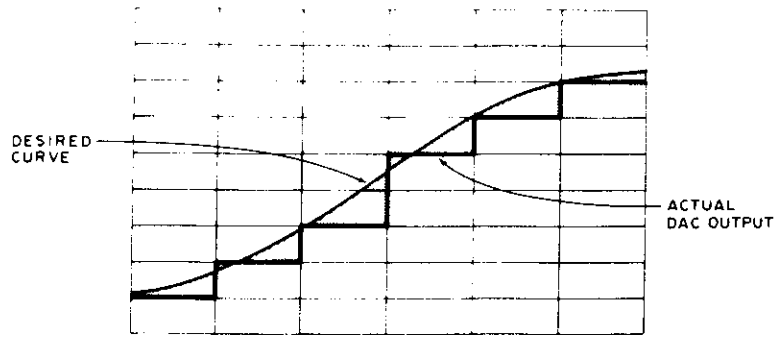


Figure 2: DAC Quantization Errors. The actual output of a computer to the analog world is a step function (in the absence of any filtering). This leads to the problem of quantization errors, depicted conceptually here by the shaded areas in between the smooth analog function and its closest step function approximation. Low precision digital to analog conversions accentuate this problem.

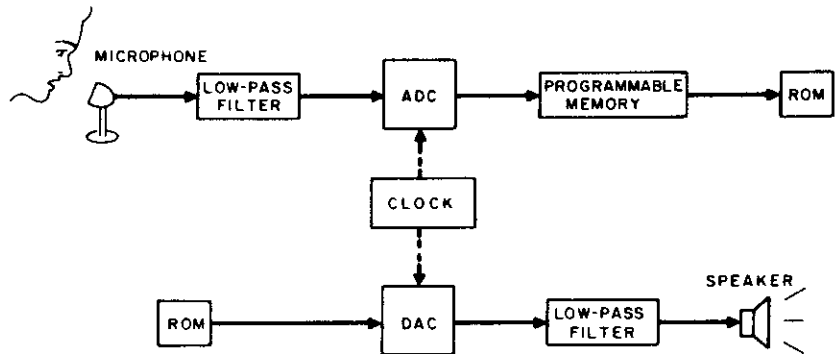


Figure 3: Waveform Playback from ROM Storage. One way to achieve a digitally controlled vocal output is to first digitize a passage of human speech, then store the digital pattern in memory. For a commercial product, such as a talking calculator, the limited vocabulary required makes this a feasible avenue of design, especially when a single mass produced ROM can be used in the final product. In an experimenter's system, the ROM is not needed, and programmable memory can be substituted during experiments. This is probably the least expensive way to augment an existing computer's capability with vocal output, but the memory requirements limit its use to small vocabularies. The quality of the result varies with the ADC (and DAC) sampling rate and precision.

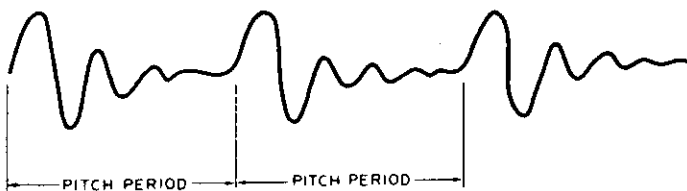


Figure 4: Typical Vowel Waveform. In principle, a vowel is a fairly long sustained passage of sound with repetitive characteristics. The vowel sounds are produced physiologically by the resonances of the vocal tract, and are controlled electronically by the formant filters which produce the equivalent of vocal tract resonances.

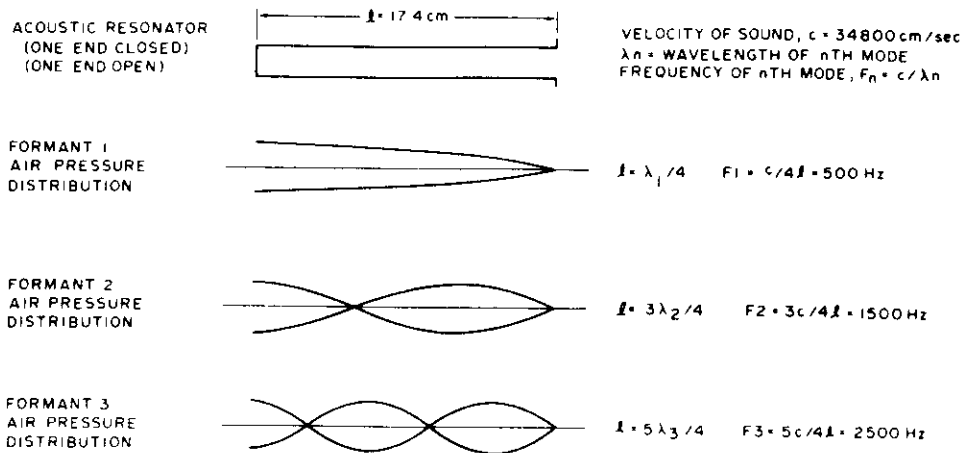


Figure 5: Tube Resonances. Temporarily ignore the complicated shape of the vocal tract and simplify it to a tube 17.4 cm long. Applying the equations of physics to acoustic waves in air gives resonances at several modes or natural frequencies. The standing waves along the tube at each frequency are shown, and identified as formant 1, formant 2 and formant 3. In the actual vocal tract, a more complicated and time varying geometry changes the resonances as a sound is created.

sweeping rapidly to a new level. It is of interest that this jitter and movement of the pitch rate has a direct effect on the perception of speech because of the harmonic structure of the speech signal. In fact, accurate and realistic modelling of the natural pitch structure is probably the one most important ingredient of good quality synthetic speech. In order to have smooth pitch changes across whole sentences, the number of separate stored waveform cycles still gets unreasonable very quickly. From these observations of the cyclic nature of vowels, let us move in for a closer look at the structure of the speech signal and explore more sophisticated possibilities for generating synthetic speech.

How Do We Talk?

The human vocal tract consists of an air filled tube about 16 to 18 cm long, together with several connected structures which make the air in the tube respond in different ways (see figure 1). The tube begins at the vocal cords, or glottis, where the flow of air up from the lungs is broken up into a series of sharp pulses of air by the vibration of the

vocal cords. Each time the glottis snaps shut, ending the driving pulse with a rapidly falling edge, the air in the tube above vibrates or rings for a few thousandths of a second. The glottis then opens and the airflow starts again, setting up conditions for the next cycle.

The length of this vibrating air column is the distance from the closed glottis up along the length of the tongue and ending at the lips, where the air vibrations are coupled to the surrounding air. If we now consider the frequency response of such a column of air, we see that it vibrates in several modes or resonant frequencies corresponding to different multiples of the acoustic quarter wavelength. There is a strong resonance or energy peak at a frequency such that the length of the tube is one quarter wavelength, another energy peak where the tube is three quarter wavelengths, and so on at every odd multiple of the quarter wavelength. If a tube 17.4 cm long had a constant diameter from bottom to top, these resonant energy peaks would have frequencies of 500 Hz, 1500 Hz, 2500 Hz and so on. These resonant energy peaks are known as the formant frequencies. Figure 5 illustrates the simple acoustic resonator and related physical equations.

The vocal tract tube, however, does not have a constant diameter from one end to the other. Since the tube does not have constant shape, the resonances are not fixed at 1000 Hz intervals as described above, but can be swept higher or lower according to the shape. When you move your tongue down to say "ah," as in figure 6, the back part is pushed back toward the walls of the

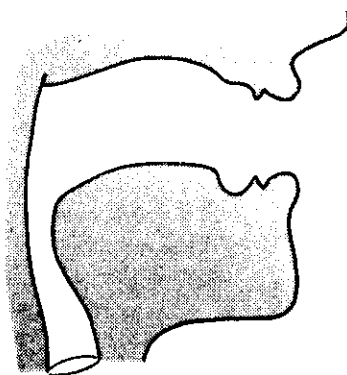


Figure 6: "ah" as in "father." In figure 1, the vocal tract was shown in schematic form. Here is a similar figure showing how the tract has been modified to produce the vowel sound "ah." The human typically closes off the nasal cavity and widens out the oral cavity by opening the mouth during this sound.

throat and in the front part of the mouth the size of the opening is increased. The effect of changing the shape of the tube in this way is to raise the frequency of the first resonance or formant 1 (F1) by several hundred Hz, while the frequency of formant 2 (F2) is lowered slightly. On the other hand, if you move your tongue forward and upward to say "ee," as in figure 7, the size of the tube at the front, just behind the teeth, is much smaller, while at the back the tongue has been pulled away from the walls of the throat, leaving a large resonant cavity in that region. This results in a sharp drop in F1 down to as low as 200 or 250 Hz, with F2 being increased to as much as 2200 or 2300 Hz.

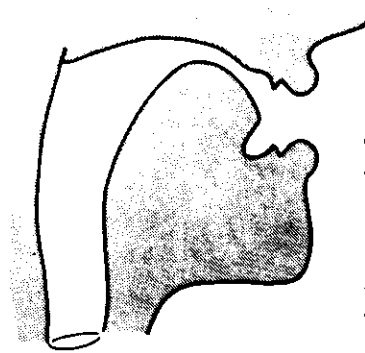


Figure 7: "ee" as in "heed." In contrast to figure 6, when the "ee" vowel sound is created, the mouth opening tends to be narrowed; and the upper end of the vocal tract is restricted. This lowers the frequency of the first resonant mode and raises the frequencies of the second and third. Referring to table 1, the "ee" vowel sound has some of the highest resonances for formants F2 and F3 and the lowest for F1.

We now have enough information to put together the circuit for the oral tract branch of a basic formant frequency synthesizer. After discussing that circuit, we will continue on in this way, describing additional properties of the speech mechanism and building up the remaining branches of the synthesizer circuit.

A Speech Synthesizer Circuit

To start with, we must have a train of driving pulses, known as the voicing source, which represents the pulses of air flowing up thru the vibrating glottis. This could be simply a rectified sine wave as in figure 8. To get different voice qualities, the circuit may be modified to generate different waveform shapes.

This glottal pulse is then fed to a sequence of resonators which represent the formant frequency resonances of the vocal tract. These could be simple operational amplifier bandpass filters which are tunable over the range of each respective formant. Figure 9 shows the concept of a typical resonator circuit which meets our requirements. IC1, IC2 and IC4 form the actual bandpass filter, while IC3 acts as a digitally controlled resistance element serving to vary



Figure 8: Voiced Sounds from the Glottis. Sounds which have definite pitch are called voiced sounds. In the natural larynx, these sounds are generated by the vocal chords and drive the vocal tract at the glottis. In an electronic analog, the voiced sounds can be generated by a programmable counter (to set the frequency) which in turn creates a sine wave of the same frequency. A rectified sine wave is a good source for the glottal pulses used in the electronic model of a larynx used in the author's approach to speech generation.

the resonant frequency of the filter. Several such resonator circuits are then combined as in figure 10 to form the vocal tract simulator. The voicing amplitude control, AV, is another digitally controlled resistance similar to IC3 of figure 9.

This gain controlled amplifier configuration is the means by which the digital computer achieves its control of speech signal elements. The data of one byte drives the switches to set the gain level of the amplifier in question. In figures 10, 13 and 15 of this article, this same variable resistance under digital control is shown symbolically as a resistor with a parameter name,

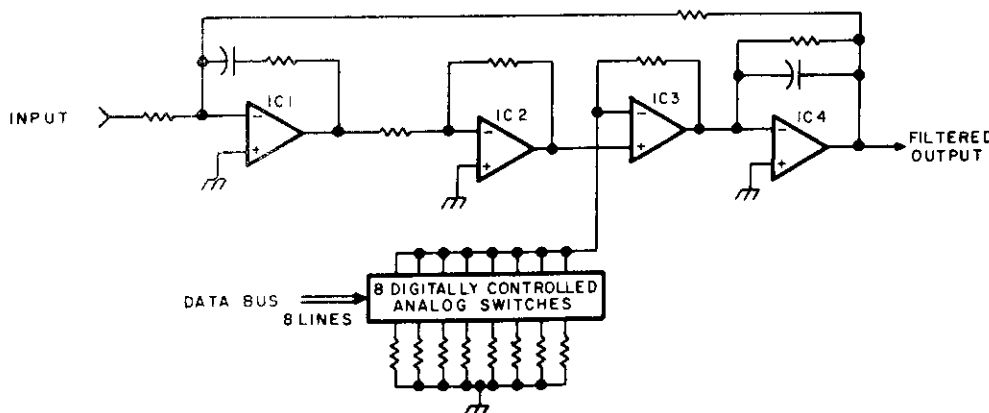


Figure 9: Typical Formant Resonator Circuit. A digitally controlled band pass filter can be built from four operational amplifiers and 8 digitally controlled analog switches. The filter characteristics are set by the choice of the resistance and capacitance elements as well as the digital control word. The operational amplifier IC3 serves as a gain controlled amplifier in the feedback loop, which alters the filter resonance.

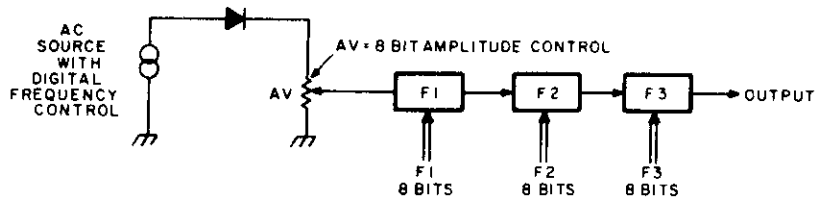


Figure 10: A first approximation of the voice synthesizer can be constructed by using three formant filters in series with differing resonance settings all controlled by 8 bit digital words. The resistance indicated as AV is an operational amplifier circuit (see IC3 of figure 9) with a digital gain control input. It is thus a programmable element of gain less than unity, in other words the electronically controlled equivalent of a variable resistance. This notation of a controlled resistance is used in figures 13 and 15 as well.

	F1	F2	F3
heed	250	2300	3000
hid	375	2150	2800
head	550	1950	2600
had	700	1800	2550
hod	775	1100	2500
paw	575	900	2450
hood	425	1000	2400
who	275	850	2400

Table 1: Steady State English Vowels. The vowel sounds are made by adjusting the formant resonances of the human vocal tract to the frequencies listed in this table. These figures are approximate, and actual formant resonances vary from individual to individual. In a speech synthesizer based upon an electronic model of the vocal tract, the formant frequencies are set digitally using operational amplifier filters with adjustable resonant peaks.

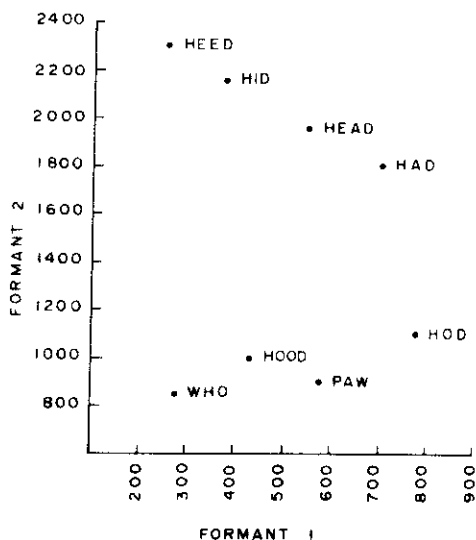


Figure 11: The Steady State English Vowels. The distinctions between various vowel sounds can be illustrated by plotting them on a two dimensional graph. The horizontal axis is the formant 1 frequency, the vertical axis is the formant 2 frequency. A location for each vowel utterance can be determined experimentally by locating the resonance peaks with an audio spectrum analyzer.

rather than as an operational amplifier with analog switches.

Generating Vowel Sounds

The vocal tract circuit as shown thus far is sufficient to generate any vowel sound in any human language (no porpoise talk, yet). Most of the vowels of American English can be produced by fixed, steady state formant frequencies as given in table 1. A common word is given to clearly identify each vowel. The formant frequency values shown here may occasionally be modified by adjacent consonants.

An alternative way to describe the formant relationships among the vowels is by plotting formant frequencies F1 vs F2 as in figure 11. F3 is not shown here because it varies only slightly for all vowels (except those with very high F2, where it is somewhat higher).

The F1-F2 plot provides a convenient space in which to study the effects of different dialects and different languages. For example, in some sections of the United States, the vowels in "hod" and "paw" are pronounced the same, just above and to the right of "paw" on the graph. Also, many people from the western states pronounce the sounds in "head" and "hid" alike, about halfway between the two points plotted for these vowels on the graph.

A few English vowels are characterized by rapid sweeps across the formant frequency

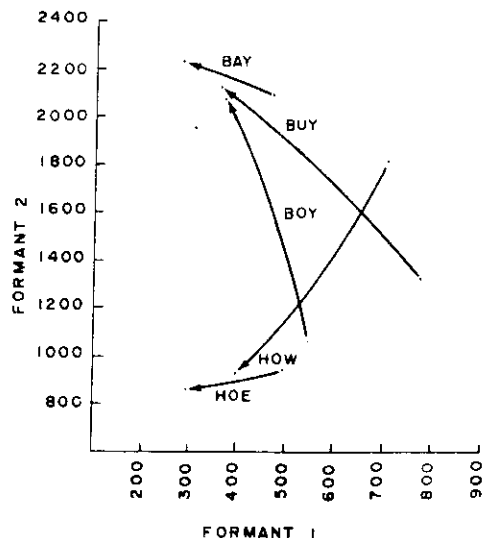


Figure 12: English Diphthongs. A diphthong is a sound which represents a smooth transition from one vowel sound to another during an utterance. The time duration of the swap from one point to another in formant space is typically 150 to 250 ms. This graph shows typical starting and ending points for several common diphthong sounds.

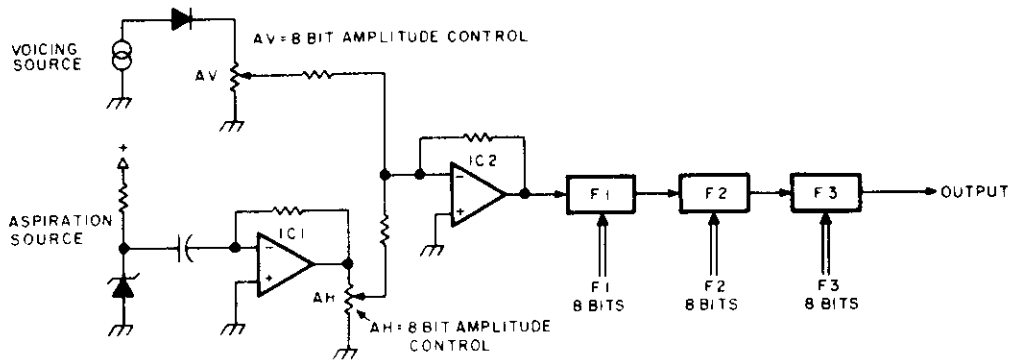


Figure 13: Synthesizer with Aspiration Noise Generator. Not all utterances are vowels. By adding a digitally controlled noise generator to the circuit of figure 10, it is possible to synthesize the consonant sounds known as "stops." In this circuit, the amplitude versus time characteristics of the noise pulse are determined by an 8 bit programmable gain control AH (shown symbolically as a resistor). The output of the noise source is mixed with the voicing source with the analog sum being routed to the formant filters. The noise generator is a zener diode.

space rather than the relatively stable positions of those given in table 1. These sweeps are produced by moving the tongue rapidly from one position to another during the production of that vowel sound. Approximate traces of the frequency sweeps of formants F1 and F2 are shown in figure 12 for the vowels in "bay," "boy," "buy," "hoe" and "how." These sweeps occur in 150 to 250 ms roughly depending on the speaking rate.

Consonant Sounds

Consonant sounds consist mostly of various pops, hisses and interruptions imposed on the vibrating column of air by the actions of several components of the vocal tract shown in figure 1. We will divide them into four classes: 1) stops, 2) liquids, 3) nasals, and 4) fricatives and affricates. Considering first the basic 'stop consonants,' "p," "t," "k," "b," "d" and "g," the air stream is closed off, or stopped, momentarily at some point along its length, either at the lips, by the tongue tip just behind the teeth or by the tongue body touching the soft palate near the velum. Stopping the air flow briefly has the effect of producing a short period of silence or near silence, followed by a pulse of noise as the burst of air rushes out of the narrow opening.

The shape of the vocal tract with the narrow opening at different points determines the spectral shape of the noise pulse as well as the formant locations when voicing is started. Both the noise burst spectrum and the rapid sweeps of formant frequency as the F1-F2 point moves into position for the following vowel are perceived as characteristic cues to the location of the tongue as the stop closure is released. We need only add a digitally controlled noise generator to the vocal tract circuit of figure 10 to simulate the noise of the burst of air at the closure release and we can then generate all the stop consonants as well as the vowels. Figure 13 shows the speech synthesizer with such a noise generator added. The breakdown noise of a zener diode is amplified by IC1 and amplitude is set by the digitally controlled resistor AH. IC2 is a mixer amplifier which combines the glottal source and aspiration

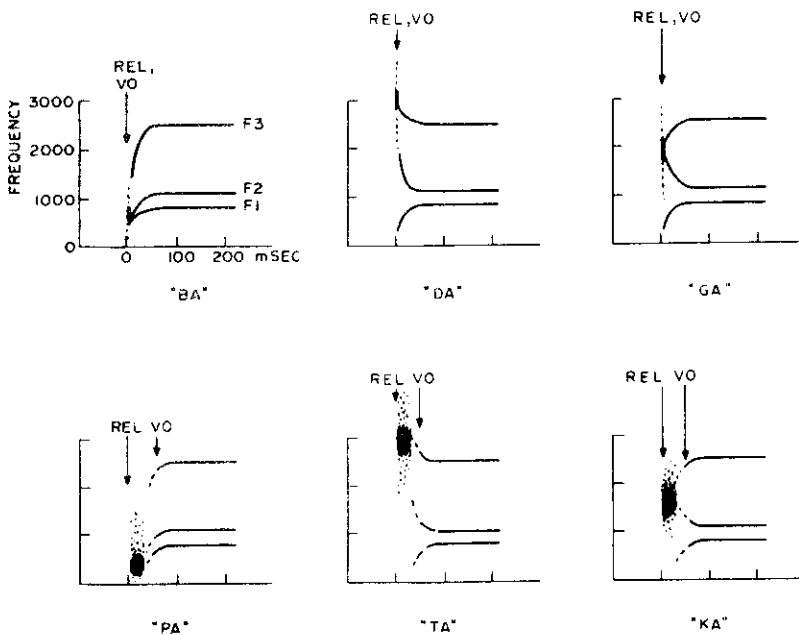


Figure 14: Stop Consonant Patterns. This figure illustrates 6 different stop consonant patterns. The release of the stop closure (start of noise pulse) is at the point marked by "REL" and the beginning of the voicing sounds is marked by "VO". Note the typical transition of the vowel formants as the steady state is reached.

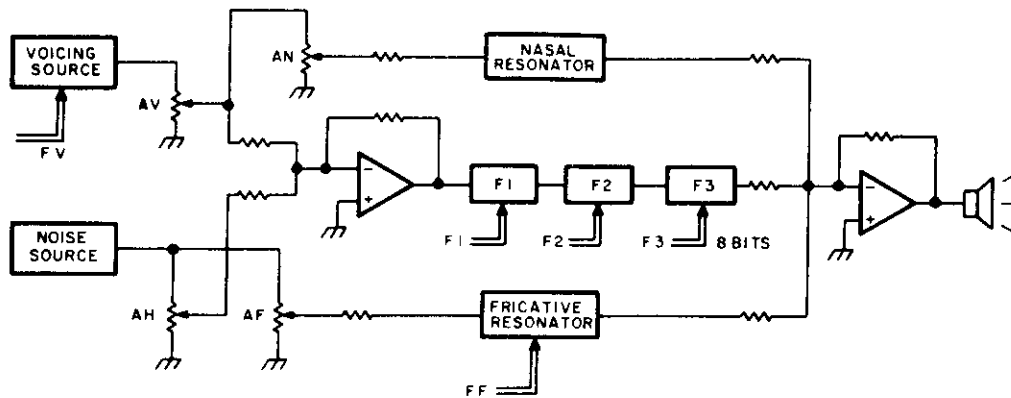


Figure 15: The Complete Synthesizer. This diagram shows the organization of a complete synthesizer which includes a wide variety of parameters. The voicing frequency and amplitude are set by parameters FV and AV. The noise pulses of stop consonants are generated with the programmable gain element AH. The fricative resonator with amplitude AF and frequency resonance FF are used to generate fricatives like "s" and "sh." The normal vowel sounds are generated by control of the formant frequencies F1, F2 and F3, and a nasal resonator with amplitude AN and fixed frequency characteristics is used to add varying amounts of nasal sounds. The result of signals processed through the nasal, formant and fricative paths is summed by a final operational amplifier and used to drive the output speaker.

noise at the input to the formant resonators.

It is important to notice at this point the range of different sounds that can be generated by small changes in the relative timing of the control parameters. The most useful of these timing details is the relationship between the pulse of aspiration noise and a sharp increase in the amplitude of voicing (see figure 14). For example, if we set the noise generator to come on for a noise pulse about 40 ms long and immediately after this pulse, F1 sweeps rapidly from 300 up to 775 Hz and F2 moves from 2000 down to 1100 Hz, the sound generated will correspond to moving the tip of the tongue down rapidly from the roof of the mouth. Observe, however, that the formant output is silent after the noise pulse until the voicing amplitude is turned up. If voicing is turned on before or during a short noise burst, the circuit generates the sound "da," whereas if the voicing comes on later, after a longer burst and during the formant frequency sweeps, the output sounds like "ta." This same timing distinction characterizes the sounds "ba" vs "pa" and "ga" vs "ka," as well as several other pairs which we will explore later. Figure 14 gives the formant frequency patterns needed to produce all the stop consonants when followed by the vowel "ah." When the consonant is followed by a different vowel, the formants must move to different positions corresponding to that vowel.

The important thing to note about a stop transition is that the starting points of the frequency sweeps correspond to the point of

closure in the vocal tract, even though these sweeps may be partially silent for the unvoiced stops "p," "t" and "k," where the voicing amplitude comes on after the sweep has begun.

The second consonant group comprises the liquids, "w," "y," "r" and "l." These sounds are actually more like vowels than any of the other consonants except that the timing of formant movements is crucial to the liquid quality. "W" and "y" can be associated with the vowels "oo" and "ee," respectively. The difference is one of timing. If the vowel "oo" is immediately followed by the vowel "ah," and then the rate of F1 and F2 transitions is increased, the result will sound like "wa." A comparison of the resulting traces of F1 and F2 vs time in "wa" with the transition pattern for "ba" in figure 14 points out a further similarity. The

	Resonator Frequency (FF)	Fricative Amplitude (AF)
sh, zh	2500	.9
s, z	5000	.7
f, v	6500	.4
th	8000	.2

Table 2: Fricative Spectra. A fricative sound typically consists of a pulse of high frequency noise. The various types of fricatives are classified according to the spectral profile of the pulse. For the electronic model described here, the fricative amplitude and resonator frequency for several sounds are listed in this table.

direction of movement is basically the same, only the rate of transition of "ba" is still faster than for "wa." Thus we see the parallelism in the acoustic signal due to the common factor of lip closeness in the three sounds "ua," "wa" and "ba." "Y" can be compared with the vowel "ee" in the same way, so the difference between "ia" and "ya" is only a matter of transition rates. Generally, "l" is marked by a brief increase of F3, while "r" is indicated by a sharp drop in F3, in many cases, almost to the level of F2.

The third group of consonants consists of the nasals, "m," "n" and "ng." These are very similar to the related voiced stops "b," "d" and "g," respectively, except for the addition of a fixed "nasal formant." This extra formant is most easily generated by an additional resonator tuned to approximately 1400 Hz and having a fairly wide bandwidth. It is only necessary to control the amplitude of this extra resonator during the "closure" period to achieve the nasal quality in the synthesizer output.

The fourth series of consonants to be described are the fricatives, "s," "sh," "z," "zh," "f," "v" and "th" and the related affricates "ch" and "j." The affricates "ch" and "j" consist of the patterns for "t" and "d" followed immediately by the fricative "sh" or "zh," respectively, that is, "ch" = "t+sh" and "j" = "d+zh." The sound "zh" is otherwise rare in English. An example occurs in the word "azure." With the letters "th," two different sounds are represented, as contained in the words "then" and "thin." All the fricatives are characterized by a pulse of high frequency noise lasting from 50 to 150 msec. The first subclassification of fricatives is according to voicing amplitude during the noise pulse, just as previously described for the stop consonants. Thus, "s," "sh," "f," "ch" and "th" as in "thin" have no voicing during the noise pulse, while "z," "zh," "v," "j" and "th" as in "then" have high voice amplitude. When a voiceless fricative is followed by a vowel, the voicing comes on during the formant sweeps to the vowel position, just as in the case of the voiceless stops. The different fricatives within each voice group are distinguished by the spectral characteristics of the fricative noise pulse. This noise signal differs from that previously described for the stop bursts in that it does not go thru the formant resonators, but is mixed directly into the output after spectral shaping by a single pole filter. Table 2 gives the fricative resonator settings needed to produce the various fricative and affricate consonants. Fricative noise amplitude settings are shown on a scale of 0 to 1.

Product Information

At the time this article goes to press, a synthesizer module incorporating several detail refinements and improvements over the circuits of this article is being developed by the author and associates. A detailed user's guide will be supplied with the Computalker module which illustrates the timing relationships needed to produce all the consonant-vowel and vowel-consonant combinations which occur in natural speech. This can serve as a reference guide for creating your speech output software which generates the proper control patterns from text inputs. Write to Computalker, 821 Pacific St No. 4, Santa Monica CA 90405 for the latest information on this module.

The Complete Synthesizer

The system level diagram of a complete synthesizer for voice outputs is summarized in figure 15. The information contained in this article should be sufficiently complete for individual readers to begin experimenting with the circuitry needed to produce speech outputs. In constructing a synthesizer on this model, the result will be a device which is controlled in real time by the following parameters:

- AV = amplitude of the voicing source, 8 bits
- FV = frequency of the voicing source, 8 bits
- AH = amplitude of the aspiration noise component, 8 bits
- AN = amplitude of the nasal resonator component, 8 bits
- AF = amplitude of the fricative noise component, 8 bits
- F1 = frequency of the formant 1 filter, 8 bit setting.
- F2 = frequency of the formant 2 filter, 8 bit setting.
- F3 = frequency of the formant 3 filter, 8 bit setting.
- FF = frequency of fricative resonator filter, 8 bit setting.

This is the basic hardware of a system to synthesize sound; in order to complete the system, a set of detailed time series for settings for these parameters must be determined (by a combination of the theory in this article and references, plus experiment with the hardware). Then, software must be written for your own computer to present the right time series of settings for each sound you want to produce. Commercial synthesizers often come with a predefined set of "phonemes" which are accessed by an appropriate binary code. The problem of creating and documenting such a set of phonemes is beyond the scope of this introductory article, but is well within the dollar and time budgets of an experimenter. ■

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Hardware & software for speech synthesis

by Lloyd Rice Computalker Consultants
P. O. Box 1951 Santa Monica, CA 90406

The process of generating voice output with a computer can be broken down into several steps. We will examine the operations at each step to determine the flow of information into and out of the step. This examination will give us the background needed to decide which parts of the overall process should be wired into a hardware device, and which parts should be kept as software to retain flexibility and control over the process. Perhaps the easiest way to carry out such an examination is by following an example phrase thru the system as it is transformed into a speech signal and sent to the loudspeaker:

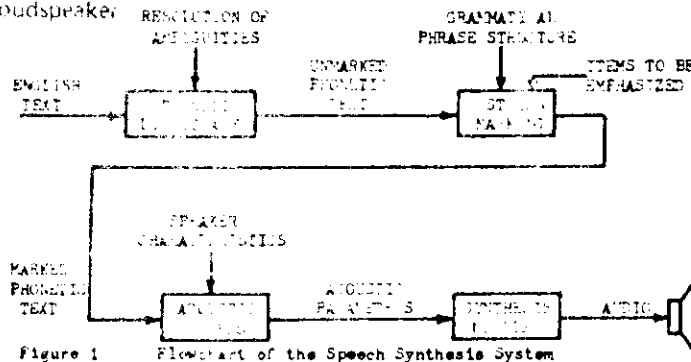


Figure 1
Flowchart of the Speech Synthesis System

Figure 1 shows a flowchart of the speech output system to be described. We will see that the kinds of external information needed for the first 2 steps is quite difficult to obtain and can require large amounts of processing, whereas the information needed in the third step is easily determined, and in most applications can be set as constants in the system. Finally, the acoustic parameters contain all the information necessary to control the last step, the actual synthesizer, to produce audio output. As a result of these observations, we will see that in most cases, one should specify the material to be synthesized in the form of marked phonetic text rather than raw English text. In order to present a more complete description, however, we will begin with the first step shown in Figure 1, input of English text.

Beginning with the sample text, "This is computer speech.", we first consult a phonetic dictionary, which performs a direct translation to phonetic text. A phonetic coding scheme suitable for this purpose which is compatible with the ASCII character set and Teletype output was developed by the Advanced Research Projects Agency (ARPA) as a part of a recent speech recognition study. That phonetic code, known as ARPABET, is listed in Table 1. The output of the phonetic dictionary in our example would be, "DHHS/IHZ/KAHMPYUWTER/SPIY2CH *".

The main problem which arises at this stage is due to homographs, words that are spelled the same but pronounced differently. Two different types of homographs, however, present quite different problems. The first type consists mainly of short words such as "bow," pronounced either as in "tied a bow" and "bow and arrow," or as in "off the starboard bow." In these cases, the pronunciation can usually be resolved by examining the surrounding context. The other type of ambiguity is a lesser noticed but very widespread phenomenon in English: the situation where a word has a different stress pattern depending on whether it is used as a noun or a verb. As an example of this, notice the difference in

"It was a dull subject," and "They were going to subject him to cruel punishment." It is not evident from the spelling which usage is intended, and requires that a fairly complete grammatical analysis be carried out to make that decision. One advantage at this point is that good use can be made of the recovered grammatical structure in the next step, where a more elaborate assignment of stress is performed.

The second step in the synthesis process deals with the assignment of sentence stress levels to the phonetic text string. To clarify that operation we will first have a closer look at the nature of the linguistic feature known as stress. The stress will be coded as a numerical value attached to a vowel in the phonetic string. That value will be realized later by the synthesizer in three different ways: as an increase in the pitch frequency, as a lengthening of the vowel duration, and to some extent as an increase in amplitude. The primary or highest level of stress is marked as a "1" following the vowel symbol. Secondary stress, marked with a "2," has less extreme acoustic effects than primary stress. As many as 3 or 4 distinct levels of stress may be marked in a sentence.

With regard to its communicative value, stress serves two quite distinct functions. The sentence stress pattern, together with timing and intonation, serves to communicate the grammatical structure to the listener. One can think of the grammatical structure as being transformed into a stress and intonation pattern by the speaker which is then decoded back into the phrase structure by the listener. Using the term "grammar," I am here including several kinds of information about words, such as the noun-verb distinction discussed above as well as syntactic information about the phrase and clause structure. The second function of the stress pattern is to indicate which item or items in a sentence are to be given special emphasis. The meaning of a sentence can be shifted around by emphasizing different items. The sources of information needed for marking these two components of the stress pattern are quite different and must be considered separately. Our example, with the stress pattern marked, would be something like, "DHHS/IHZ/KAHMPYUWTER/SPIY2CH *". Notice also that the word and utterance boundary markers have been kept explicitly in the text string.

The purpose of the portion of the system described thus far is simply to generate strings of phonetic text with marked stress patterns which are to be synthesized by the 2 steps in the bottom row of Figure 1. Marked phonetic text strings can be obtained in other ways, of course. In the case of predetermined phrases, marked phonetic strings can be stored instead of raw English text, making the synthesis task much simpler. On the other hand, consider synthesis of speech from an information network of some kind. The grammatical information could come from a phrase structure grammar which is being driven by relationships in the network. Items in the network would be coded as phonetic strings, or in essence, references to the phonetic dictionary described above. There are many significant problems remaining with this approach, but it is perhaps one of the more exciting applications of synthetic speech. The third box in the flowchart in Figure 1 is the acoustic rules section. In order to describe what the acoustic rules are and what they do, we must first look at the acoustic structure of speech. The speech code must be broken down into components so it can be synthesized by controlling, in real time, a limited number of parameter values. To a good

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approximation, speech can be represented by the model shown in Figure 2.

This model requires 9 parameter control values consisting of 5 frequency controls and 4 amplitude controls. The box labeled "pulse source" is a controllable frequency oscillator which is adjusted dynamically to determine the voice

pitch. The boxes labeled "resonator" are tunable, single-pole, bandpass resonators which determine the frequency or spectral shape of the speech signal in different ways. The data bus symbol used to represent the control inputs indicates that each parameter can be controlled by at most 8 bits from the computer's output bus. The data rates needed to control the

Table 1		Computer Representation		Example	Phoneme	Computer Representation		Example
COMPUTER	PHONETIC	1-Character	2-Characters			1-Character	2-Characters	
	i	I	IY	beat	p	P	P	pet
	I	I	IH	kit	t	t	T	ten
	e	E	EY	bait	k	k	K	kit
	e	E	EH	bet	b	b	B	bet
	æ	A	AE	bat	d	d	D	debt
	a	A	AA	Bob	g	g	G	get
	A	A	AH	hut	h	h	HH	hat
	o	O	AO	bought	f	f	F	fat
	o	O	OW	boat	θ	T	TH	thing
	u	U	UH	book	s	s	S	sof
	u	U	UW	boot	s or /	S	SH	shut
	ə	X	AX	about	v	v	V	vat
	ɜ	X	IX	rosea	θ	D	DH	that
	ɜ	R	ER	bird	s	z	Z	zoo
	aU or əw	W	AW	down	z or ʒ	Z	ZH	azure
	aI or əy	Y	AY	buy	ʒ or ʒ	C	CH	church
	ɔI or ɔy	O	OY	boy	ʒ	J	JH	judge
	y	Y	Y	you	ʌ	H	WH	which
	w	W	W	wit	syl l,l	L	EL	battle
	r	R	R	rent	syl m,m	M	EM	bottom
	l	L	L	let	syl n,n	N	EN	button
	m	M	M	net	flapped t,f	F	DX	batter
	n	N	N	net	glottal stop	Q	Q	
	o	G	GX	sing	silence	-	-	
				non-speech segment		I	I	laugh, etc.
AUXILIARY SYMBOLS (1- AND 2-CHARACTER CODES ARE IDENTICAL)								
Symbol	Meaning			Symbol	Meaning			
+	Morpheme boundary			:3 or .	Fall-rise or non-term juncture			
/	Word boundary			* **	Comment (anything except * or **)			
#	Utterance boundary			' '	Apos-surround special symbol in comment			
:	Tone group boundary			()	Phoneme class information			
:1 or .	Falling or decl. juncture			< >	Phonetic or allophonic escape			
:2 or ?	Rising or inter. juncture							
STRESS REPRESENTATIONS (IF PRESENT, MUST IMMEDIATELY FOLLOW THE VOWEL)								
Value	Stress Assignment			Value	Stress Assignment			
o	No stress			3	Tertiary stress			
1	Primary stress			.	(Etc.)			
2	Secondary stress			:				

parameters are quite low, the highest rate needed for any parameter being less than 100 new settings per second.

I will not go into detail here describing the actual parameter values needed to represent particular speech sounds. An article to appear in the August, 1976 issue of *Byte Magazine* goes into some detail on the nature of the different kinds of speech sounds and how they can be generated by controlling the parameter values in such a model. Such information would, of course, be necessary to write a software implementation of the acoustic rules. For our present purposes, we consider the 9 control values as outlined above to represent an acoustic parameter model of speech. We can now turn to a discussion of the acoustic rules and the tasks they must perform to generate controls for this model.

Each phoneme, as encoded in the phonetic text string, is a symbol representing one or more acoustic speech segments, each such segment being produced by a particular pattern of values on the control parameters. Each pattern, or configuration of control values, must be held for a specific length of time before changing to the next pattern. As a first approximation then, the rules would consist of a series of table look-ups to convert each phoneme into a sequence of parameter patterns, along with the duration each pattern is to be held.

Now comes the catch! This first approximation makes rather poor speech. The problem is that the transitions between parameter values are often more important than the actual values at any given time. The flow of parameter values must be more carefully orchestrated. Actually, the only tough problem here is that correct transitions *between* phonemes are just as important as having the correct temporal structure *within* a phoneme. This means that phonemes cannot be coded as independent sets of parameter time functions which are merely joined together sequentially, but that some interaction must take place between phoneme patterns before they are sent out to the synthesizer module. Briefly, the different phonemes of a language can be classified according to the effects of boundary interactions. The transition of each parameter value across a given phoneme boundary can then be determined from the boundary characteristics of each of the neighboring phonemes. Such boundary behavior information can be stored in phoneme look-up tables.

In addition to assigning initial parameter values and mapping the transitions across boundaries, the acoustic rules must also assign and modify durations. For example, a stressed vowel is given a longer duration than the same vowel in an unstressed position.

A third function the acoustic rules must perform—probably the most important for natural sounding speech—is to assign the time pattern of values to the pitch frequency parameter. First, an archetypal intonation pattern is chosen on the basis of punctuation (retained in the phonetic text just for this purpose). A period selects a falling pitch, a comma signals a level pause, and a question mark indicates a rising pattern. Other diacritical marks could be defined in the phonetic string to generate more complex pitch patterns such as singing. The selected archetypal pitch pattern is then modified locally by specific phonemes. Such local modification of the pitch pattern is one of the effects of a stress level marked on a vowel. Also, some consonants affect the pitch value slightly.

To complete the synthesis process, the acoustic parameters generated by the acoustic rules are output, in real time, to a synthesizer module such as sketched in Figure 2 and

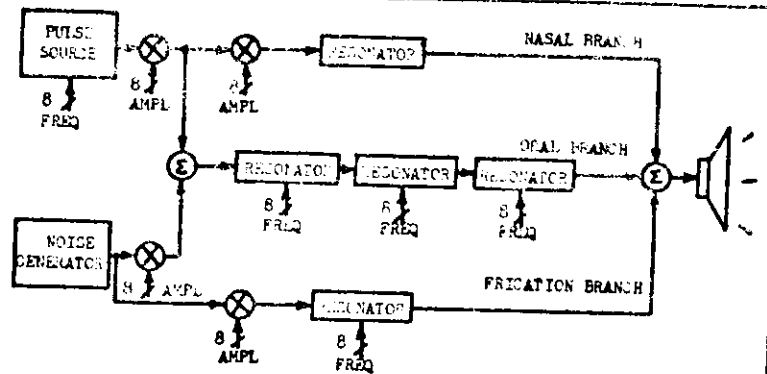


Figure 2 A Model of the Structure of Speech

described in the forthcoming *Byte* article. The synthesizer constructs an audio frequency signal as specified by the control parameters. The audio signal is then sent to a loud-speaker as the speech output.

It would be impractical to consider simulating the synthesizer module in software because of the speed needed to generate speech in real time. That task is much more appropriately handled by analog hardware. Such a hardware synthesizer module is currently being developed by Computalker, P. O. Box 1951, Santa Monica CA 90406. The Computalker synthesizer module would be driven by the microcomputer output data bus as described above. The software interface consists either of a direct, manually-controlled parameter pattern generator or an implementation of the acoustic rules. Software for the acoustic rules will also be developed by Computalker as the hardware becomes available.

I believe it is important to consider at this point some of the trade-offs involved in implementing the acoustic rules in software rather than hardware. A synthesizer system such as the Votrax VS-6 contains a hardware implementation of the basic acoustic rules. As a result, the language available for coding the phonetic text is fixed and cannot be extended. In addition, the phoneme table values are fixed so that each phoneme has a set phonetic quality. By implementing these rules in software you could retain the flexibility over pitch patterns and speech rate and also have control over the phonetic qualities which determine the language and dialect. The acoustic rules determine a number of qualities in the resulting speech which are characteristic of a particular speaker, such as the sex and age, and other qualities which vary from occasion to occasion, such as voice quality, speaking rate, distinctness of articulation, etc. Because of time constraints, a software version of the acoustic rules may not have time to handle all these possibilities as on-line variables. Of course, it is cheaper to produce a synthesizer module if a hardware acoustic rules system is not included.

How could speech output from a microcomputer be used? Several applications come to mind for the hobbyist environment, such as responses in games, voice readout of measurement data, system status warnings, etc., etc. Other applications might include telephone answering and intrusion warnings. What about generating audio tape labels automatically? Each of these applications makes its own demands for quality, naturalness and range of vocabulary needed. I would very much like to hear of your interest in computer speech output. What applications do you have in mind? What problems do you foresee? A note to the above address will assure that you receive further information as it becomes available.



CTMON, The Computalker Control Monitor

CTMON is a combined speech data editor and CT-1 playback monitor. The CT-1 control parameters are displayed as columns of decimal values on the video monitor. The speech parameter data can be read from paper tape or audio cassette, edited as desired, and written back out onto audio cassette. The speech data in the buffer may be played out to the CT-1 Synthesizer beginning at any desired data frame and played for any desired number of frames. The rate of playback may be varied, resulting in fast speech at up to 10 times the normal speaking rate, or slow speech, drawing the words out to more than 5 times the normal length of time.

VIDEO DISPLAY

CTMON is currently assembled to use a Polymorphic Systems VIDEO display board and can easily be patched to use a Processor Technology VDM-1 display system. It would not be easy to modify the program as it now stands to be able to use a TVT or Teletype-replacement type of video display. The reason for this is that I wanted to be able to scroll up or down thru the frames of data quickly to be able to move around in the data as fast as you could hit the "N" or "B" keys (for Next or Back). I did not want to take the time to rewrite the entire screen for each scroll. A scrolling feature such as the Processor Technology VDM-1 would not be of much help either, because I wanted to keep several lines fixed at the top and bottom of the screen. I usually run my video board (Polymorphic) with the width control set back so there are about 40 chars across the screen. As a result, I wrote the frame display code so it would work with less than 64 characters width (being too short sighted, tho, I did not quite get it in 32). This detail could be changed by modifying the subtraction and sign test code just before the label CT2A at location 0186H and similarly in the subroutines DISFRM at 057BH and ARROW at 05B8H. Also the column headings should then be changed. The text lines FRMTX, FORTX, and AMPTX should then be combined into one line at FRMTX. This would affect the character count at 013BH.

The program was assembled using Polymorphic's suggested video buffer address of 8800H. To change this to the address of your video display requires changing three locations as follows. If the symbol VIDBUF is the page number of the first video location, then the full 16 bit address of that location will be VIDBUF*256. With a 1K byte video buffer, VIDBUF+4 will be the next page beyond the end of video memory and VIDBUF/4 is just the bit pattern of VIDBUF shifted right 2 bits. The examples shown below would address the screen properly for VIDBUF EQU 0E0H.

<u>location</u>	<u>now contains</u>	<u>change to</u>	<u>symbol</u>
0694H	88H	0E0H	VIDBUF
069AH	8CH	0E4H	VIDBUF+4
06EDH	22H	38H	VIDBUF/4

The Polymorphic VIDEO board requires that the most significant bit be set high to get text mode instead of graphics. With a VDM-1, this will give either positive or negative characters depending on the setting of the on-board switch. If you must have the most significant bit low for your board, this may be done by changing the following locations as shown.

<u>location</u>	<u>now contains</u>	<u>change to</u>
01BAH	9BH	0DH
01D2H	0A0H	20H
059DH	80H	00
05CDH	80H	00
0681H	80H	00
0696H	0A0H	20H
06ACH	0A0H	20H
06FCH	80H	00
0737H	80H	00
073FH	80H	00

Of these, location 01BAH contains the symbol to be displayed as a left-pointer just to the right of the selected parameter data value. The Polymorphic board with a 6571 character generator ROM has a back-arrow at code 9BH. The VDM-1 which I tried this system on had a suitable character at code 0DH. You may have to put a different code here depending on your character generator ROM.

OTHER I/O

Reading and writing of speech parameter data is done under the control of separate monitor commands for each device.

<u>command</u>	<u>action</u>
TR	Read data from paper tape & replace current buffer contents
TA	Read data from paper tape & append it to data currently in buffer
TW	Punch buffer contents on paper tape (not yet implemented)
TY	List buffer contents on TTY or printer
CR	Read data from cassette & replace current buffer contents
CA	Read data from cassette & append it to data currently in buffer
CW	Write buffer contents onto cassette

Code is currently written for Teletype input (using MITS definitions) and input and output using Tarbell cassette definitions. All instructions specific to an I/O device are located near the end of the listing at locations 074FH thru 07B4H. These routines are all reasonably well commented as far as which way flags are set and what's being returned, etc. In general, it will only be necessary to change the OUT and IN addresses to those needed for your device assignments. In most of these routines, space has been left for a NOP or CMA before the bit-masking, depending on whether your device has active high or active low status.

GETC Waits for strobe from console keyboard, then gets 1 ASCII char and returns with ms bit low in A register.

KEYTST This only checks whether a key has been hit, it does not get the character. This leaves the strobe still active for GETC. If a key was hit, KEYTST returns with a non-zero condition.

TTYOUT This has not yet been coded.
 It is called from location 035CH on the command "TY", which would produce a formatted listing as opposed to a parameter data format dump to paper tape.

Two Teletype/paper tape output command processors have not been completed. TTYLIS would construct a formatted listing as mentioned above, and TPDATA would punch the buffer contents on tape in parameter data format. For Teletype operation, these would both call TTYOUT, and would call separate subroutines if a separate paper tape punch were available.

- PTRCLR** This initializes the paper tape reader at the beginning of each input transmission. Both this and PTRIN below are currently set up for a MITS 2S10 board.
- PTRIN** This subroutine waits for the paper tape reader ready flag to go on, or a time limit to expire, and if ready, gets 1 data byte from the reader and returns it in A. Note that this is 8 bit data, not ASCII, so the ms bit should not be masked off. A timer is included so you don't hang forever on an unready ready flag. If the flag does not go on within approx. 1/2 second, the routine returns zero condition. If the data byte was read normally, it returns non-zero condition with the byte in A. Registers BC are used, but restored before return.

CASSETTE I/O

The cassette routines are assembled using Don Tarbell's standard definitions of 6EH (status) and 6FH (data).

- CASCLR** This initializes the cassette reader board and clears the B register for use as a checksum accumulator. It is called once at the beginning of each input transmission, on either command "CR" or "CA". Note that the cassette should be moving forward reading leader at the time either of these commands is given.
- CASIN** This subroutine reads 1 byte from cassette. It is called with the read address in the HL registers and puts the byte just read directly into memory at (HL). The address in HL is then incremented. In addition, the value of the byte read is added to the contents of the B register, which is accumulating the checksum for the record. All registers and flags except the B register are restored before return.
- CASLDR** This sets up everything for a cassette write transfer including leader and all sync bytes. My Tarbell board is wired with a relay which closes by setting status bit 0 to a 1. This connects the audio to the cassette input. If your system is set up so that starts the cassette motor, too, you may want to add more than the 2 second delay I used. I should have cleared B here also, but that's done back in CWDATA where this is called.
- CASOUT** This writes 1 byte from A onto the cassette. Note that it does not write directly from memory as with CASIN. CASOUT also accumulates a checksum in the B register.

**BEFORE USING THE TARBELL I/O ROUTINES,
CHANGE THE FOLLOWING LOCATION:**

<u>LOC</u>	<u>NOW CONTAINS</u>	<u>CHANGE TO</u>
<u>03C7</u>	81H	96H

CASSETTE I/O Modifications for MITS ACR

The following modifications should be patched into CTMON in order to use MITS ACR cassette, assuming MITS standard port assignments, flags=06, data=07. When using this cassette I/O to write a record, the drive should be connected in RECORD mode for several seconds before typing the "CW" command. When reading a cassette data record, enter the command "CR" or "CA" while the head is reading this leader.

<u>location</u>	<u>now contains</u>	<u>change to</u>	<u>symbolic code</u>
03A8	CD 81 07	CD 96 07	CALL CASCLR2
03C6	CD 81 07	CD A5 07	CALL CASLDR
03E0	AF	C9	RET
0787	DB 6E	DB 06	IN CASF
0789	E6 10	E6 01	ANI 01
078E	DB 6F	DB 07	IN CASD
0796	3E 01	D3 06	CASCLR2 OUT CASF
0798	D3 6E	DB 07	IN CASD * reset UART
079A	CD 53 06	21 81 07	LXI H,CASCLR *unused loc
079D	3E 3C CD	CD 86 07	CALL CASIN *read & ignore
07A0	A7 07	06 00	MVI B,0 start byte
07A2	3E	C9	RET
07A5	A7 07	3E FF	MVI A,0FFH * write start byte
07A8	DB 6E	DB 06	IN CASF
07AA	E6 20	E6 80	ANI 80H
07B0	D3 6F	D3 07	OUT CASD

PLAYING and EDITING SPEECH DATA with CTMON

When you first bring up the Control Monitor system, you see frame 1 displayed across the screen with Ø's shown in each data field. Put the demonstration data tape 'HELLO' in the paper tape reader and then type the command 'TR'. Now start the tape reader. At the end of the tape, the command TR should disappear from the screen and several columns of data should be in view, beginning with frame 1. The total length of data in the buffer is shown to be 746 frames.

TYPE 'P' TO HEAR IT

With the CT-1 Synthesizer board in place and connected to an audio amplifier, you should now be able to type the command 'P' and hear the demonstration phrase. The playback sequence can be interrupted any time in the middle of the phrase by typing either Rubout or Escape. These keys serve as the general abort function for any playback operation or for falsely entered commands. Up to this point, the commands described had no arguments and were executed immediately upon typing the command letter(s). We now consider the second type of command, having a numerical argument.

One of the two kinds of timing delays controlled by CTMON is the time between frame updates to the synthesizer. This time is normally set at 10 milliseconds and can be changed using the command 'H'. Type 'H', followed by a 5 (for 5 msec), and then a carriage return. Now type 'P' again. This time the playback is at twice the normal speaking rate, with data frames being updated to the synthesizer every 5 msec. Also try H20 and again P, playing back the speech at half speed. This is not like changing the speed of a tape recording because the frequencies of the voice do not change here, only the rate of speaking changes. Remember that any command with an argument must be ended with a return.

The second kind of delay is the length of time between repeats. Type the command 'R'. You will hear the entire phrase and then after about 1 second pause, the phrase will repeat. This will continue until you hit any other key (like rubout or escape). By typing a 'W' followed by the desired wait time in msec, and return, you can change the length of time between repeats up to 65 seconds.

Using the 'S' (Start at frame) and 'L' (Length to play) commands, you can control the section of data to be played over the synthesizer. Enter the commands S600 return, and L146 return. Now when you type P you hear only the phrase "8080 microcomputer". To find where a certain sound is in the data, enter a length of approximately 50 frames and then change the starting frame by 100's until you hear the piece you want. Sometimes you will hear a thump if the playback starts or ends in the middle of a vowel. That is because of the unusually rapid onset of amplitude to the synthesizer.

MAKE IT SOUND DIFFERENT

Watch the screen as you type the command "AH". You will see the "current parameter" pointer move over to column AH, the amplitude of aspiration parameter. The display should still be positioned at frame 1. If it is not, move to frame 1 by typing the move command M1 return. Observe the column of values beginning at frame 1; 10, 20, 30, 40, ..., and locate this column of data in the listing of the demonstration phrase "HELLO". The values greater than 0 in the AH column represent the amplitude of the hissing sound produced during the h sound, frames 1 thru 8. Set the playback pointers to include just the word "hello", that is, S1 return, L50 return. Now listen carefully to the word "hello" and observe the hissing sound of the h.

You can change the value of the current parameter at the current frame by typing the desired value and a return. Note that typing only a return is the same as 0 return. This is for convenience since 0 is by far the most frequent value you will want to enter, but it is also a bit risky because stray returns seem to pop up every now and then. This would zero a location in your data. Entering a new data value into the buffer has one other effect: it moves you forward to the next frame. Now type a series of returns, at least 8, until you see all 0's in the AH column. Again play the word "ello" and notice the Cockney accent with the missing "h". Move back to frame 1, either by typing M1 return or a series of B's. Then reenter a new set of amplitude values, say in the range 80 to 100, to make it much louder. Notice that there is a fairly critical threshold of amplitude where it becomes too loud and no longer sounds like an h in the word, but just sounds like a hissing noise separated from the sound of the word itself. A smooth decay over 3 or 4 frames instead of a "square corner" also has quite a lot to do with this disassociation effect.

Now experiment with the pitch control, F0. Type "F0" to select that parameter and then go down the column from frame 1 setting each frame's F0 value to a constant, say 60. An easy way to do this is by typing a 60 in the usual way for the first frame and then typing the ditto command (") until you have set the desired number of frames. Continue this thru at least frame 44. Now listen to "hello" and hear that it has lost all intonation and is pronounced in a monotone. Experiment with different pitch values and try constructing a new contour for the F0 parameter. The original F0 control for "hello" has a peak around frame 12. Try moving the major peak down to about frame 35. You may find it helpful to sketch the curve you want on graph paper and read off the new values to be entered.

Look around thru the data and try changing other parts to see what effect it has. For example, thru most of the entire demonstration phrase, flattening out the F3 control doesn't make very much difference as long as it's set roughly to a mid-range value. However, if you reduce the value of F3 at frames 583-593, the "r" sound in "your" will disappear. Notice the peaks of F3 at frame 508 in "standard" and at frame 689 in "micro". This is typical of the "r" sound.

The CTMON Control Monitor software system is supplied with the CT-1 Synthesizer board both as ASCII source and assembled and punched in Intel Hex object format. The code was assembled using a modified version* of the popular Microtec/Processor Tech/IMSAI assembler. One characteristic of that system that I am not particularly happy with is that every line must begin with its line number. That takes a lot of paper tape. Also, the lines are formatted nicely with a lot of spaces. Even with the comments in, the actual source code would be about half that much paper if fields were marked with just a single space or tab character. For the most part the syntax and symbols used are those in common use elsewhere. Some things which may be undefined in other assemblers are DT for define text, definition of the full symbols PSW and SP, and the use of multiplication, division and modulus operators in operand expressions.

At the time this software was written, I did not have the facilities to assemble it on my own machine, and had the job done at several different places. That is my excuse for the missing parts and its generally incomplete state of development. In one sense it's a makeshift system anyway, because of the lack of a cheap high-resolution graphics capability, which would allow the editing to be done graphically instead of numerically. Such an editor would be much easier to work with. In any case, I will fully support this software system, which means that I will be sending out update packages to all purchasers. That will include the Insert and Delete command processors in the near future, and of course, patches for any bugs discovered as more people begin using the system.

*PDS-1, modified by Steve Zook, at the Computer Store, Santa Monica, CA

COMPUTALKER CONTROL MONITOR -- COMMAND SUMMARY
 FOR VERSION 1.10 DEC 13, 76

INDICATES A DECIMAL NUMBER 0 THRU 65535
 <CR> INDICATES A CARRIAGE RETURN

#<CR> CHANGE VALUE OF CURRENT PARAMETER TO # AND
 STEP FORWARD ONE FRAME
 " REPEAT LAST #<CR> COMMAND WITH SAME #
 AF OPEN AF FOR EDITING
 AH OPEN AH FOR EDITING
 AN OPEN AN FOR EDITING
 AV OPEN AV FOR EDITING
 B STEP BACK ONE FRAME
 CA APPEND DATA TO BUFFER FROM CASSETTE
 CR READ FROM CASSETTE & REPLACE BUFFER
 CN WRITE BUFFER CONTENTS ONTO CASSETTE
 D#<CR> DELETE # FRAMES, CURRENT FRAME THRU CURRENT+#-1 *
 F0 OPEN F0 FOR EDITING
 F1 OPEN F1 FOR EDITING
 F2 OPEN F2 FOR EDITING
 F3 OPEN F3 FOR EDITING
 FF OPEN FF FOR EDITING
 H#<CR> HANG # MSEC BETWEEN FRAMES DURING PLAY
 I#<CR> INSERT # FRAMES BEFORE CURRENT FRAME *
 (CURRENT FRAME IS DUPLICATED # TIMES)
 KILL KILL THE ENTIRE BUFFER CONTENTS
 L#<CR> SET TO PLAY # FRAMES ONLY
 M#<CR> MOVE TO FRAME #
 N STEP FORWARD ONE FRAME
 P PLAY
 R REPEAT PLAY
 S#<CR> SETUP TO START PLAY AT FRAME #
 TA APPEND TO BUFFER FROM PAPER TAPE
 TR READ PAPER TAPE & REPLACE BUFFER CONTENTS
 TW WRITE BUFFER CONTENTS TO PAPER TAPE (PUNCH) *
 (WILL BE TP IN NEXT VERSION)
 TY LIST BUFFER CONTENTS ON TTY *
 (WILL BE TL IN NEXT VERSION)
 W#<CR> WAIT # MSEC BETWEEN REPEAT PLAYS
 X EXIT TO USER DEFINED LOCATION *

IN THE NEXT VERSION, THE SYNTAX OF # COMMANDS WILL BE
 CHANGED TO #M INSTEAD OF THE PRESENT M#<CR>. THE
 COMMAND WILL BE EXECUTED IMMEDIATELY ON TYPING THE
 COMMAND LETTER AND WILL NOT WAIT FOR THE <CR>.

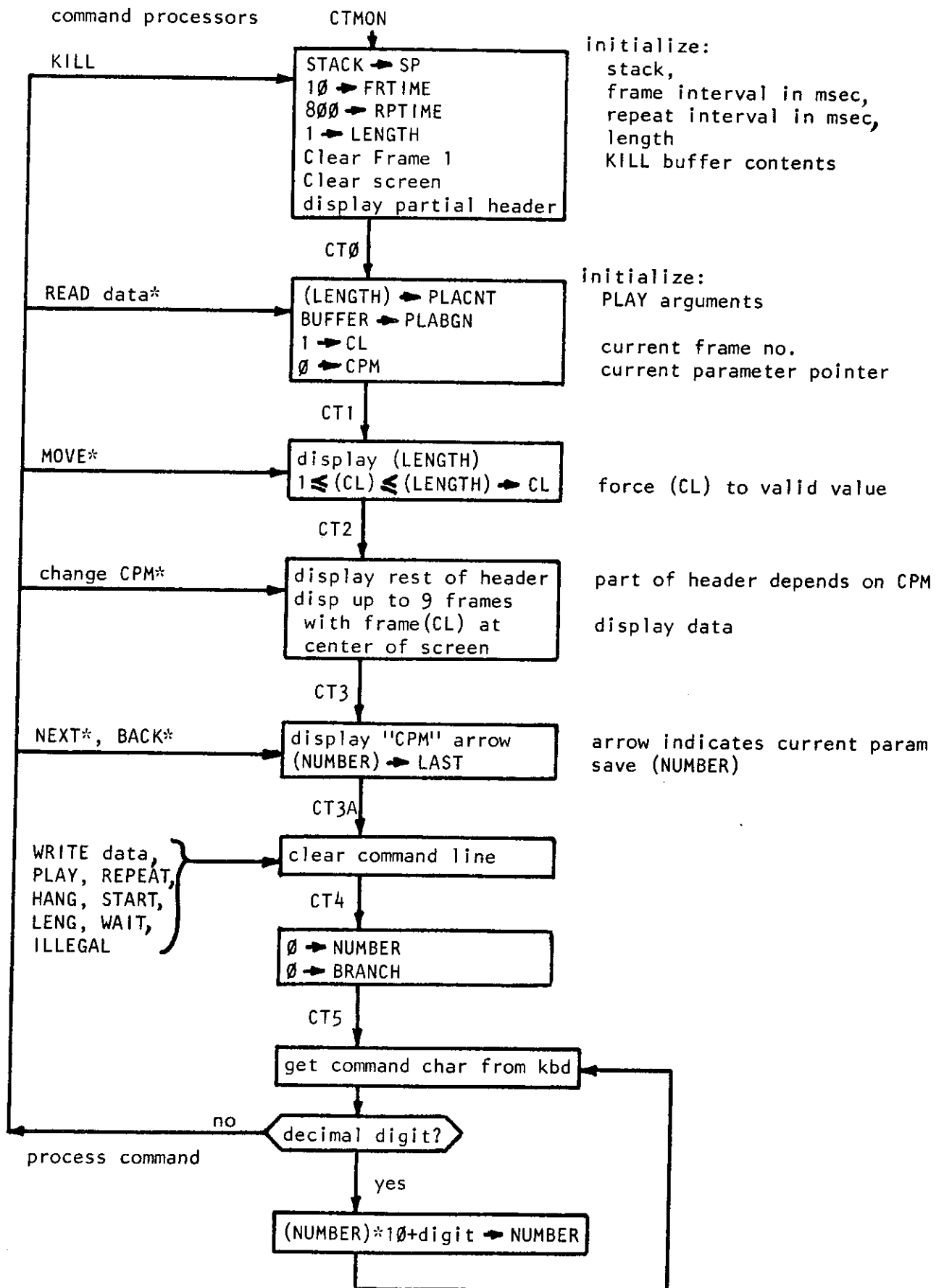
* not implemented
 in this version

The Vector Graphics FLASHWRITER is another memory-mapped video system that CTMON will work with. The board uses memory locations D000H thru D3FFH. To patch the standard version of CTMON assembled at address 0100H, make the following changes:

<u>location</u>	<u>now contains</u>	<u>change to</u>
01BAH	9BH	0DH
01D2H	A0H	20H
059DH	80H	00H
05CDH	80H	00H
0681H	80H	00H
0694H	88H	D0H
0696H	A0H	20H
069AH	8CH	D4H
06ACH	A0H	20H
06EDH	22H	34H
06FCH	80H	00H
0737H	80H	00H
073FH	80H	00H

To patch the special version of CTMON assembled for Micropolis systems at address 2A00H, make the following changes:

<u>location</u>	<u>now contains</u>	<u>change to</u>
2BBAH	9BH	0DH
2BD2H	A0H	20H
2F9DH	80H	00H
2FCDH	80H	00H
3081H	80H	00H
3094H	88H	D0H
3096H	A0H	20H
309AH	8CH	D4H
30ACH	A0H	20H
30EDH	22H	34H
30FCH	80H	00H
3137H	80H	00H
313FH	80H	00H



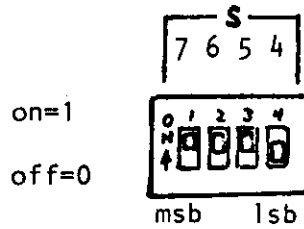
* must clear "CPM" arrow
before executing command

"HOW THE CT-1 INTERFACE WORKS"

During the period of time that Addresses A4-A7 compare with the setting of the DIP-Switch, and both $\overline{\text{PWR}}$ and SOUT are active, the data D0-D7 is strobed into an 8 bit register; also the Address A0-A3 is then strobed into a 4 bit latch. As the data bits are stored, they are also passed to the 8 bit Digital-to-Analog Converter (DAC). When $\overline{\text{PWR}}$ or SOUT go inactive, the voltage from the DAC is strobed into one of 14 sample-and-holds, the destination to be determined by the contents of the 4 bit address latch. Nine of these sample-and-holds are used to control the CT-1 parameters listed below, one is used to turn the CT-1 audio output (which appears on a RCA connector at the top edge of the board) on and off, and 4 are reserved for future additions.

<u>Address</u> A3---A0	<u>Channel</u> (hex)	<u>Mnemonic</u>	<u>Parameter</u>
0 0 0 0	0	AV	Voicing Amplitude
0 0 0 1	1	F0	Voicing Frequency
0 0 1 0	2	F1	Formant 1 Frequency
0 0 1 1	3	F2	Formant 2 Frequency
0 1 0 0	4	F3	Formant 3 Frequency
0 1 0 1	5	AH	Aspiration Amplitude
0 1 1 0	6	AF	Frication Amplitude
0 1 1 1	7	FF	Frication Frequency
1 0 0 0	8	AN	Nasal Amplitude
1 0 0 1	9	}	Reserved for Future Expansion
1 0 1 0	A		
1 0 1 1	B		
1 1 0 0	C		
1 1 0 1	D		not used
1 1 1 0	E		not used
1 1 1 1	F	SW	Audio On/Off Switch

Orientation of the output port selector switch
 (for serial numbers 031 and up)



Output port numbers are the above selected hex digit followed by the channel number. Hex channel numbers are as follows:

<u>Channel</u>	<u>Mnemonic</u>	<u>Description</u>
0	AV	Amplitude of Voicing
1	F0	Frequency of Voicing (fundamental)
2	F1	Formant 1 Frequency
3	F2	Formant 2 Frequency
4	F3	Formant 3 Frequency
5	AH	Amplitude of Aspiration (hiss)
6	AF	Amplitude of Frication
7	FF	Frequency of Frication
8	AN	Amplitude of Nasal
9		} reserved for future expansion
A		
B		
C		
D		
E		
F	SW	Audio on-off Switch

Your Model CT-1 was shipped with the selector switch set to accept data on ports E0 - EF, as shown above. The Control Monitor software system has been assembled to output data on the same ports. To modify the code to output data on another set of ports, redefine the variable CTBASE to the desired address of the parameter AV.

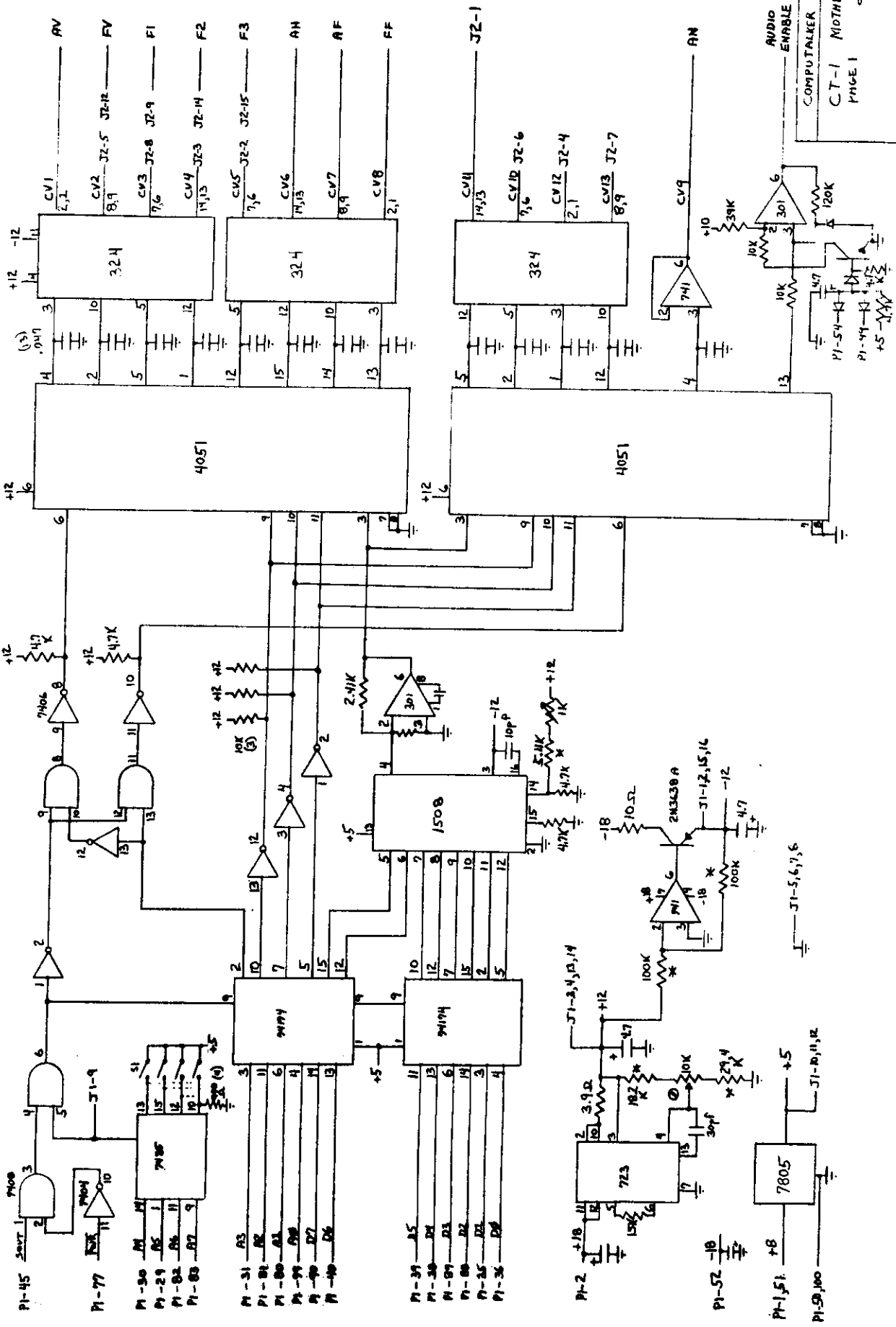
COMPUTALKER CT-1 BUS REQUIREMENTS

Following are the 8800-type (S-100) Bus signals used by the CT-1 Speech Synthesizer:

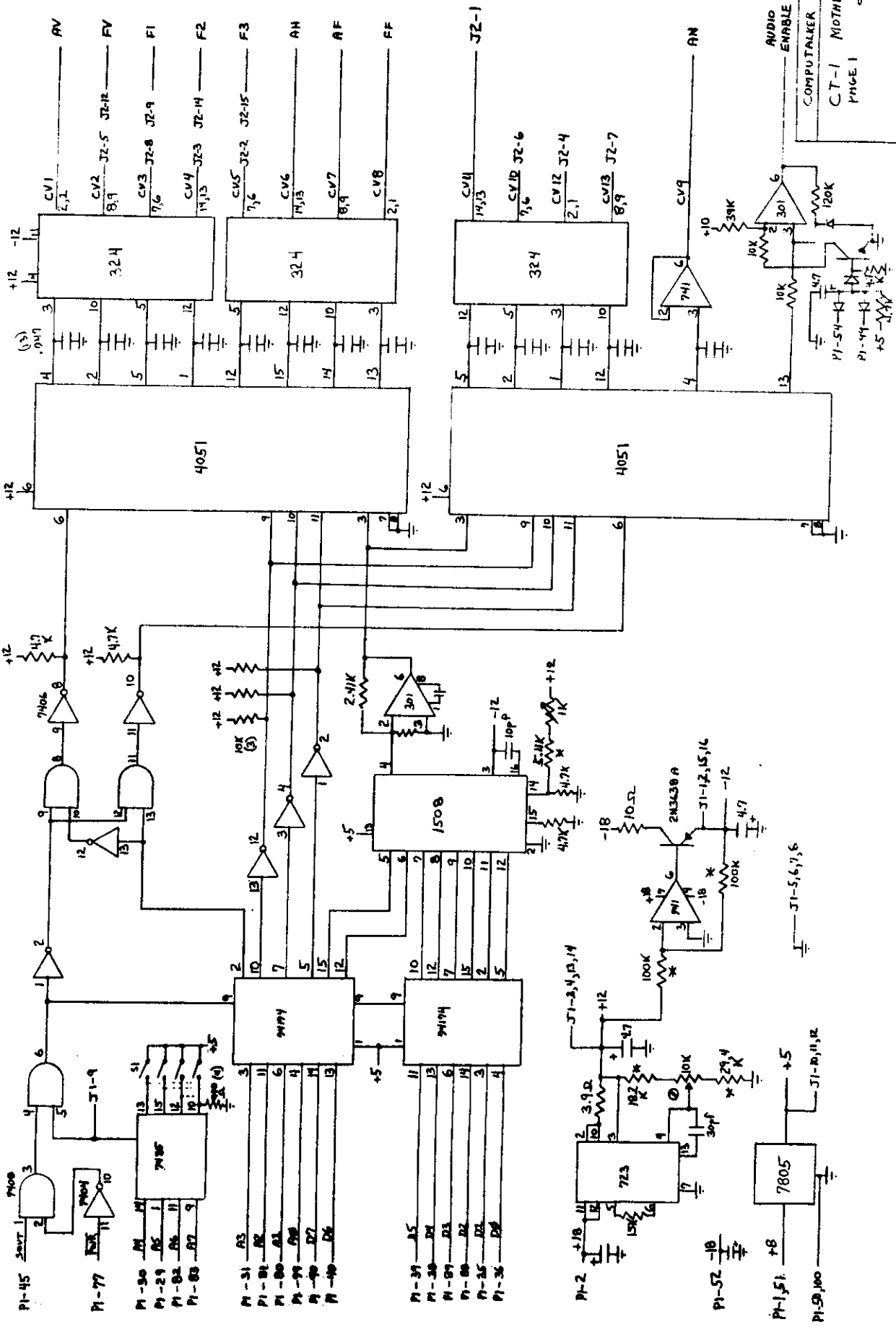
<u>PIN</u>	<u>NAME</u>	<u>ACTIVE</u>	<u>DESCRIPTION</u>	
36	D0	HIGH	LSB	
35	D1	"	} 8 bit parameter data from computer to CT-1.	
88	D2	"		
89	D3	"		
38	D4	"		
39	D5	"		
40	D6	"		
90	D7	"		MSB
79	A0	HIGH		LSB
80	A1	"	} 4 bit address from computer selects 1 of 16 possible parameter outputs to be updated.	
81	A2	"		
31	A3	"		MSB
30	A4	HIGH		LSB
29	A5	"	} 4 bit address from computer selects which block of 16 output ports is used. This address is compared against the on-board DIP switch to enable transfer of data.	
82	A6	"		
83	A7	"		MSB
77	$\overline{\text{PWR}}$	LOW		} These two signals in active state allow transfer of data from com- puter to CT-1.
45	SOUT	HIGH		

COMPUTALKER CT-1 BUS REQUIREMENTS (cont.)

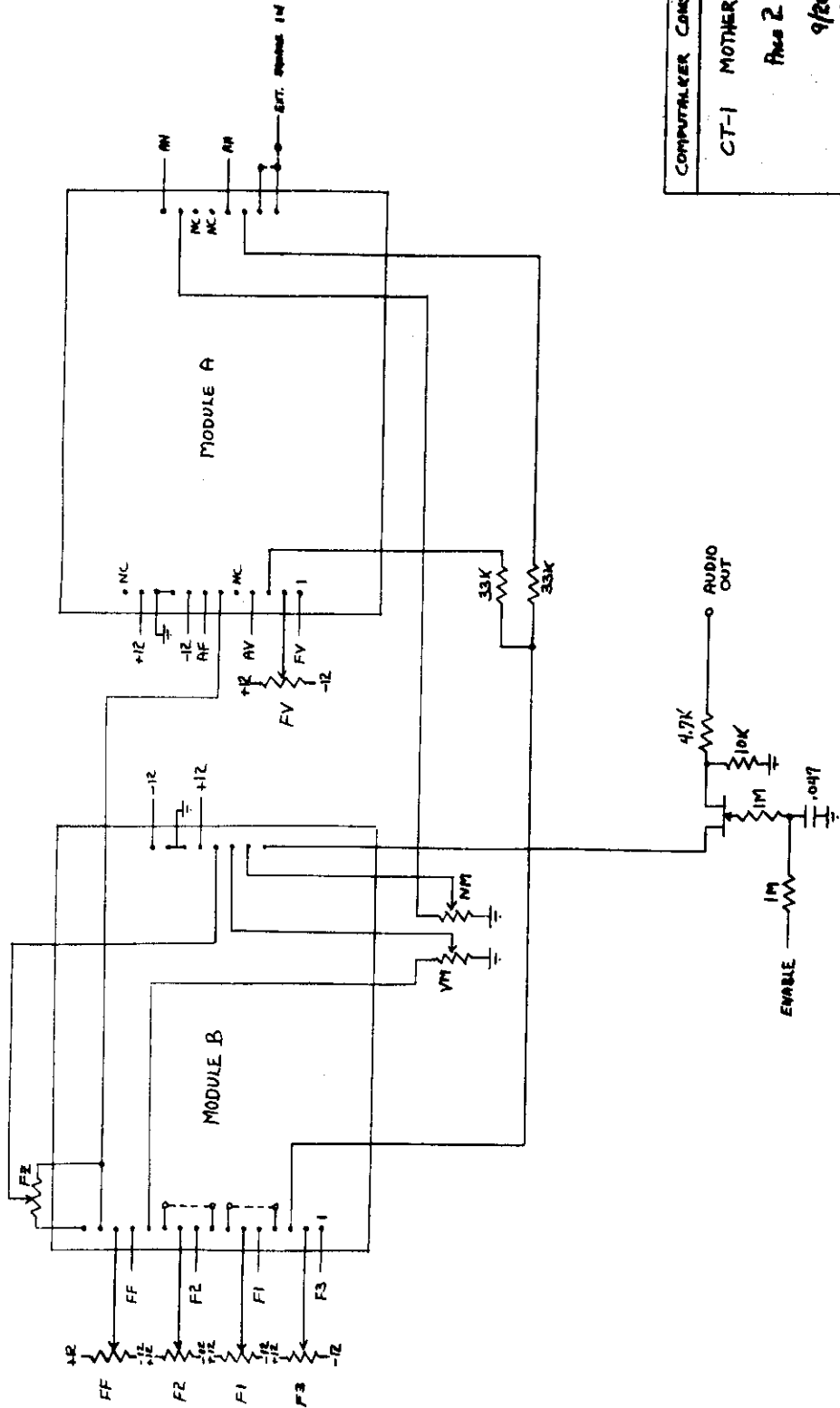
<u>PIN</u>	<u>NAME</u>	<u>ACTIVE</u>	<u>DESCRIPTION</u>
54	$\overline{\text{EXT CLR}}$	LOW	} When either of these signals is brought to its active level, the on-board audio switch is turned off.
99	$\overline{\text{POC}}$	LOW	
1,51	+8V	}	These three un-regulated voltages are supplied by the computer (or external supply) to the CT-1. If desired, +5 Volts may be used instead of +8V by a simple change on the board.
50,100	Ground		
2	+16V		
52	-16V		



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Five of the nine synthesizer control parameters have a frequency-related control function. These are Frequency of Voicing (F0), Frequency of Formant 1 (F1), Frequency of Formant 2 (F2), Frequency of Formant 3 (F3), and Frequency of Frication (FF). Of these, Frequency of Voicing (F0) moves as a direct exponential function of the control value, while the other four, F1, F2, F3, and FF, move as an inverse exponential function of their control values.

If C represents the control value as sent on the computer bus for one of these frequency control parameters, and F represents the resulting frequency in hz, the following relations hold between C and F. To compute the frequency as a function of the control value, use the equation

$$F = A * \text{EXP}(B * C)$$

and to determine the control value needed to produce a desired frequency, use the equation

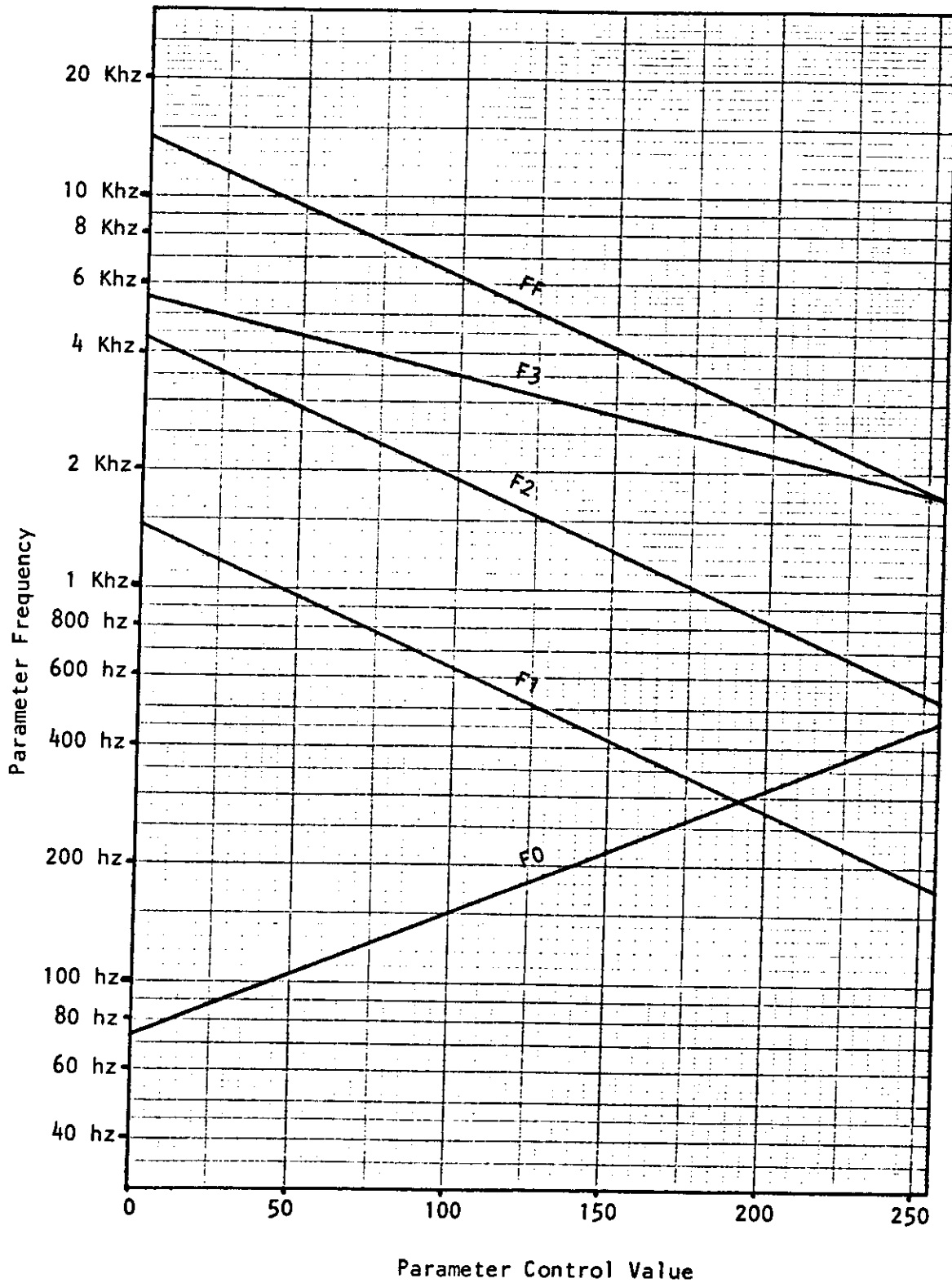
$$C = \frac{\text{LOG}_e(F/A)}{B}$$

Values for the constants A and B for the five parameters are:

	A	B
F0	73.4	.00722
F1	1452	-.0083
F2	4356	-.0083
F3	5508	-.0046
FF	14160	-.0083

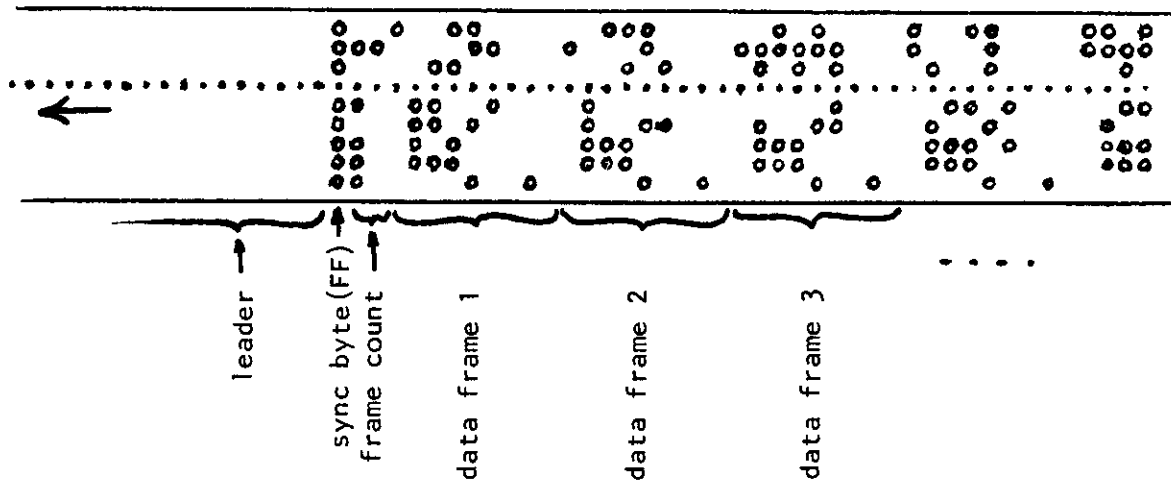
These conversion functions are also given in graphical form on the next page.

Frequency Conversion for Parameters F0, F1, F2, F3, and FF



The CT-1 speech parameter data format

Paper tape parameter data punched for or by the Control Monitor, CTMON, has the following format:



The frame count is punched low order byte first. In this example, the frame count is 02EA hex, or 746 decimal frames.

Each data frame consists of 9 bytes with the parameter values in the order AV, F0, F1, F2, F3, AH, AF, FF, and AN. There is no restriction on the value of any parameter within the range 0-255.

DATA RATE:

Within each frame, the 9 individual parameters should be sent out to the CT-1 using the appropriate OUT address. At least 20 microseconds should be allowed between individual parameter updates. This time is needed for the sample-and-hold capacitor to be fully charged to the new value in the worst case of a sweep to the opposite extreme value.

The data is coded so that the time between frame refreshes for a normal speech rate is 10 milliseconds. The CT-1 board may be updated with new frame data at any desired rate, of course, to vary the rate of speech. At an update rate faster than approx. 2 msec per frame, the speech loses intelligibility. The maximum length of time between frame updates is approx. 50 to 100 milliseconds, at which time the sample-and-hold capacitors cannot hold the value steady.

phonetic
symbol

"HELLO"

FRAME	AV	F0	F1	F2	F3	PH	PF	FF	AN
1	1	120	92	101	147	10	0	120	0
2	2	120	97	101	147	20	0	120	0
3	2	116	99	102	147	30	0	120	0
4	<i>h</i> 3	116	104	100	146	40	0	120	0 H
5	3	115	110	107	151	40	0	120	0
6	11	115	115	110	155	30	0	120	0
7	76	116	107	115	150	20	0	120	0
8	117	116	101	120	150	10	0	120	0
9	149	119	102	127	160	0	0	120	0
10	159	123	104	137	159	0	0	120	0
11	155	124	109	149	150	0	0	120	0
12	146	124	119	159	155	0	0	120	0
13	138	123	132	170	159	0	0	120	0
14	127	113	142	176	162	0	0	120	0
15	<i>E</i> 101	101	147	162	164	0	0	120	0 e
16	122	65	141	179	162	0	0	120	0
17	149	76	119	174	160	0	0	120	0
18	167	64	110	167	157	0	0	120	0
19	175	59	104	165	150	0	0	120	0
20	170	56	101	160	159	0	0	120	0
21	179	53	99	150	160	0	0	120	0
22	179	51	98	150	164	0	0	120	0
23	170	46	97	150	167	0	0	120	0
24	177	46	90	157	165	0	0	120	0
25	176	46	101	160	167	0	0	120	0 l
26	<i>l</i> 174	46	103	163	166	0	0	120	0
27	172	46	105	165	166	0	0	120	0 l
28	169	46	106	166	164	0	0	120	0
29	167	49	106	171	165	0	0	120	0
30	164	50	110	175	166	0	0	120	0
31	163	52	112	180	164	0	0	120	0
32	162	54	114	182	164	0	0	120	0
33	162	58	117	183	162	0	0	120	0
34	162	63	119	184	163	0	0	120	0
35	162	70	121	185	161	0	0	120	0
36	161	75	123	185	159	0	0	120	0
37	<i>o</i> 150	60	124	184	150	0	0	120	0
38	152	67	120	184	150	0	0	120	0
39	141	93	132	183	160	0	0	120	0 o
40	106	96	139	183	163	0	0	120	0
41	<i>w</i> 76	98	151	183	166	0	0	120	0
42	57	100	164	182	173	0	0	120	0
43	29	100	174	180	177	0	0	120	0
44	12	101	180	176	176	0	0	120	0
45	0	102	183	176	176	0	0	120	0
46	0	102	187	175	176	0	0	120	0
47	0	102	188	173	176	0	0	120	0
48	0	101	189	171	176	0	0	120	0
49	0	100	186	170	176	0	0	120	0
50	0	99	185	167	176	0	0	120	0
51	0	98	182	165	176	0	0	120	0
52	0	96	180	163	176	0	0	120	0
53	0	93	177	162	176	0	0	120	0
54	0	91	172	161	176	0	0	120	0
55	0	89	169	160	176	0	0	120	0

00	0	86	165	159	178	0	0	128	0
01	0	84	161	158	178	0	0	128	0
02	0	80	155	157	178	0	0	128	0
03	0	77	150	156	178	0	0	128	0
04	0	73	145	156	178	0	0	128	0
05	0	69	138	156	177	0	0	128	0
06	0	66	132	155	176	0	0	128	0
07	0	63	126	155	174	0	0	128	0
08	0	61	119	155	172	0	0	128	0
09	0	58	113	155	170	0	0	128	0
10	0	55	106	155	167	0	0	128	0
11	0	52	99	155	163	0	0	128	0
12	0	49	91	155	159	0	0	128	0
13	0	46	85	155	154	0	0	128	0
14	0	44	81	155	148	0	0	128	0
15	0	40	74	155	144	0	0	128	0
16	0	39	68	155	140	0	0	128	0
17	0	37	65	155	137	0	0	128	0
18	0	37	62	157	135	0	0	128	0
19	14	42	62	157	135	0	0	128	0
20	56	56	61	156	135	0	0	128	0
21	83	54	64	156	136	0	0	128	0
22	118	62	62	154	141	0	0	128	0
23	156	71	61	151	143	0	0	128	0
24	183	80	62	147	146	0	0	128	0
25	208	87	68	141	147	0	0	128	0
26	238	96	68	137	149	0	0	128	0
27	212	95	62	132	151	0	0	128	0
28	214	108	63	125	153	0	0	128	0
29	214	104	69	117	160	0	0	128	0
30	213	107	67	111	158	0	0	128	0
31	211	111	101	107	154	0	0	128	0
32	205	114	125	104	152	0	0	128	0
33	194	116	142	102	154	0	0	128	0
34	176	118	160	101	160	0	0	128	0
35	146	119	177	100	166	0	0	128	0
36	137	118	176	98	166	0	0	128	26
37	133	115	177	96	168	0	0	128	61
38	127	114	177	94	170	0	0	128	91
39	125	114	176	91	172	0	0	128	106
40	122	114	176	89	173	0	0	128	111
41	120	114	176	87	174	0	0	128	114
42	116	113	176	85	175	0	0	128	114
43	110	112	176	83	176	0	0	128	115
44	101	109	161	81	182	0	0	128	112
45	76	96	164	80	187	0	0	128	102
46	53	84	156	80	193	0	0	128	89
47	21	95	224	85	198	0	0	150	57
48	0	101	214	93	200	0	0	200	24
49	0	104	193	103	199	0	0	250	0
50	0	108	178	116	196	0	0	250	0
51	REL → 34	110	142	122	191	35	40	250	0
52	0	112	123	127	166	30	30	250	0
53	0	113	111	131	160	30	20	250	0
54	0	113	97	135	175	30	10	250	0
55	0	113	86	139	172	30	0	250	0
56	13	111	78	140	167	20	0	250	0
57	68	111	73	142	165	0	0	250	0

114	198	114	78	145	159	0	0	258	0
115	198	117	66	147	155	0	0	268	0
116	198	120	62	149	152	0	0	280	0
117	199	122	66	151	146	0	0	280	0
118	196	124	76	152	146	0	0	128	0
119	198	127	87	146	149	0	0	128	0
120	184	129	109	148	146	0	0	128	0
121	129	139	137	156	149	0	0	128	0
122	76	142	163	154	168	0	0	128	0
123	98	144	167	153	172	0	0	128	0
124	94	144	162	152	165	0	0	128	0
125	98	141	173	157	152	0	0	128	0
126	71	128	138	159	169	0	0	128	0
127	86	117	228	148	161	0	0	128	0
128	4	116	215	155	168	0	0	128	0
129	17	112	187	128	153	0	0	128	0
130	31	189	99	121	149	30	30	128	0
131	29	184	88	112	147	25	0	128	0
132	18	181	86	186	146	20	0	128	0
133	5	97	183	188	151	15	0	128	0
134	92	91	168	98	159	18	0	128	0
135	116	81	153	188	161	0	0	128	0
136	155	77	154	185	166	0	0	128	0
137	141	76	156	189	178	0	0	128	0
138	113	73	157	112	173	0	0	128	0
139	92	78	165	115	168	0	0	128	0
140	15	71	214	114	163	0	0	128	0
141	7	78	217	113	168	0	0	128	0
142	2	69	176	118	158	0	0	128	0
143	1	68	79	189	156	0	0	128	0
144	36	67	183	118	168	38	38	128	0
145	22	67	118	111	166	25	0	128	0
146	12	67	189	116	172	28	0	128	0
147	17	68	55	121	178	15	0	128	0
148	15	67	72	131	182	18	0	128	0
149	53	65	131	141	182	0	0	128	0
150	76	65	187	152	179	0	0	128	0
151	138	63	97	168	177	0	0	128	0
152	228	68	93	166	175	0	0	128	0
153	227	58	94	178	172	0	0	128	0
154	213	57	98	174	169	0	0	128	0
155	213	54	88	175	173	0	0	128	0
156	288	51	98	174	175	0	0	128	0
157	184	48	94	173	177	0	0	128	0
158	149	43	188	169	188	0	0	128	0
159	99	48	185	169	184	0	0	128	0
160	88	38	111	165	185	0	0	128	0
161	47	37	123	163	187	0	0	128	0
162	17	35	145	162	189	0	0	128	0
163	9	34	158	159	185	0	0	128	0
164	2	33	157	157	179	0	0	128	0
165	1	31	147	151	179	0	0	128	0
166	1	29	158	144	168	0	0	128	0
167	1	26	188	136	163	0	0	128	0
168	1	23	88	114	169	0	0	128	0
169	44	22	88	114	192	38	38	128	0
170	11	19	141	113	215	25	0	128	0
171	35	19	142	115	222	28	0	128	0

2399		73	38	85	147	164	0	0	128	0
2400		88	38	88	143	163	0	0	128	0
2401		113	34	83	148	161	0	0	128	0
2402		113	34	83	137	168	0	0	128	0
2403		188	38	112	134	159	0	0	188	0
2404		88	81	124	128	158	0	0	88	0
2405		39	82	142	121	158	0	28	88	0
2406		17	89	84	114	158	0	38	88	0
2407		0	72	88	113	158	0	38	88	0
2408		0	78	84	189	155	0	38	88	0
2409		0	82	98	188	154	0	38	88	0
2410		0	87	77	183	153	0	38	88	0
2411		0	91	87	182	154	0	38	88	0
2412		0	93	88	188	154	0	48	88	0
2413		0	98	82	99	152	0	48	88	0
2414		0	184	74	99	154	0	48	88	0
2415		0	188	75	188	153	0	48	88	0
2416		0	112	83	185	152	0	38	88	0
2417		0	117	81	189	152	0	28	88	0
2418		0	121	82	119	153	0	0	88	0
2419		0	123	84	139	154	0	0	88	0
2420		0	138	79	142	155	0	0	128	0
2421		0	134	88	142	158	0	0	128	0
2422		0	138	88	142	159	0	0	128	0
2423		0	138	98	139	161	0	0	128	0
2424		0	139	121	124	164	0	0	128	0
2425		0	139	188	127	163	0	38	128	0
2426		178	139	172	79	148	0	0	128	0
2427		188	142	189	71	137	0	0	128	0
2428		198	144	188	88	138	0	0	128	0
2429		192	148	194	88	138	0	0	128	0
2430		193	158	204	83	134	0	0	128	0
2431		193	151	212	84	143	0	0	128	0
2432		189	151	214	88	139	0	0	128	0
2433		177	151	218	78	139	0	0	128	0
2434		158	158	215	72	138	0	0	128	0
2435		81	151	225	79	138	0	0	138	0
2436		47	158	248	81	133	0	0	138	0
2437		29	149	247	82	138	0	0	138	0
2438		11	147	132	84	128	48	38	138	0
2439		3	148	127	88	148	48	38	138	0
2440		0	142	112	88	139	48	38	138	0
2441		0	139	188	91	144	48	38	138	0
2442		0	137	184	94	147	48	38	138	0
2443		0	134	182	97	147	48	38	138	0
2444		0	132	182	98	147	38	38	128	0
2445		0	138	182	188	144	38	48	188	0
2446		0	123	182	187	144	88	48	188	0
2447		0	127	188	185	143	88	48	188	0
2448		0	125	113	185	138	88	48	188	0
2449		0	124	122	118	147	88	48	188	0
2450		3	124	127	188	145	88	48	188	0
2451		18	125	178	188	148	88	38	188	0
2452		188	125	187	189	153	0	0	188	0
2453		223	125	183	188	152	0	0	188	0
2454		219	125	158	188	151	0	0	128	0
2455		288	124	158	187	145	0	0	128	38
2456		194	122	159	118	143	0	0	128	48

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$\frac{p}{RSL}$

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288		178	122	167	113	148	0	0	128	0
289	n	171	118	189	113	152	0	0	128	100
290		132	114	186	113	152	0	0	128	100
291		118	181	194	111	153	0	0	128	100
292		85	182	200	189	155	0	0	128	0
293		54	184	220	188	158	0	0	200	0
294		25	185	219	187	155	0	0	200	0
295		0	186	182	189	158	0	0	200	0
296		0	185	86	112	144	0	0	200	0
297	o	0	184	116	117	148	0	0	200	0
298		0	182	97	114	142	0	0	200	0
299		18	98	151	115	145	0	0	200	0
300		98	91	149	122	159	0	0	200	0
301		153	81	147	123	159	0	0	188	0
302	a	152	73	152	123	156	0	0	188	0
303		136	67	154	127	158	0	0	188	0
304		91	57	153	125	158	0	0	188	0
305		15	57	158	116	153	20	40	188	0
306		2	58	153	115	141	50	50	188	0
307		0	68	187	113	137	50	50	188	0
308		0	82	84	114	143	50	50	188	0
309	s	0	83	88	112	144	50	50	188	0
310		0	87	98	112	149	20	50	188	0
311		0	87	86	113	151	10	40	188	0
312		0	75	68	113	154	0	50	188	0
313		0	83	81	115	155	0	20	188	0
314		18	85	149	139	161	0	0	188	0
315		188	88	134	148	163	0	0	188	0
316	a	178	74	113	141	165	0	0	188	0
317		178	67	183	143	167	0	0	188	0
318		228	68	86	148	171	0	0	128	0
319		243	58	74	158	188	0	0	128	0
320		253	57	75	146	166	0	0	128	0
321		255	55	74	147	162	0	0	128	0
322		227	55	73	145	163	0	0	128	0
323		212	54	73	141	163	0	0	128	0
324	r	187	51	73	135	165	0	0	128	0
325		165	49	75	129	169	0	0	128	0
326		141	46	79	126	174	0	0	128	0
327		115	43	86	117	175	0	0	128	0
328		182	39	98	118	171	0	0	128	0
329		97	38	115	115	168	0	0	128	0
330		91	39	129	113	162	0	0	128	0
331		83	35	132	113	165	0	0	128	0
332		78	33	148	115	169	0	0	128	0
333		72	25	154	114	172	0	0	188	0
334		55	17	164	118	175	0	0	188	0
335		29	11	188	121	178	0	0	188	0
336		88	8	188	128	179	20	20	188	0
337		88	13	198	121	178	50	40	188	0
338		88	22	185	128	174	50	40	188	0
339		88	29	177	119	178	50	40	188	0
340	z	88	35	177	119	167	50	40	188	0
341		88	48	178	128	186	50	40	188	0
342		88	48	185	128	189	20	50	188	0
343		56	39	165	121	174	0	50	188	0
344		44	35	157	123	188	0	0	188	0
345		52	32	155	124	186	0	0	188	0

346		38	27	138	127	248	0	0	188	0
347		38	27	132	128	232	0	0	128	0
348		34	27	151	158	200	0	0	128	0
349	d	89	27	157	158	200	0	0	128	e
350		87	23	182	131	200	0	0	128	0
351		38	22	187	132	200	0	0	128	r
352		47	21	181	128	200	0	0	128	0
353		40	28	188	120	200	0	0	128	0
354		28	28	182	119	247	0	0	128	0
355		38	17	204	114	231	0	0	128	0
356		22	17	216	178	174	0	0	128	0
357		28	17	214	187	166	0	0	128	0
358		18	22	211	118	164	0	0	128	0
359		10	28	287	189	158	0	0	128	0
360		12	32	288	119	159	0	0	128	0
361		14	41	212	118	159	0	0	128	0
362		12	49	198	117	168	0	0	128	0
363	d	22	32	181	112	168	0	0	128	d
364	$REL \rightarrow$	29	34	172	189	157	58	78	128	0
365		37	33	173	113	157	0	0	128	0
366		82	37	169	119	159	0	0	128	0
367	z	93	36	168	121	182	0	0	128	0
368		79	34	169	123	180	0	0	128	e
369		33	49	173	124	189	0	20	128	0
370		27	49	188	124	189	0	30	128	0
371		13	33	218	124	178	0	88	128	0
372	z	18	88	248	123	189	0	88	128	s
373		18	89	219	128	188	0	88	128	0
374		18	78	179	117	188	0	88	128	0
375		22	81	158	119	184	0	58	128	0
376		19	84	130	188	157	0	28	128	0
377		131	83	125	130	159	0	0	128	0
378		168	84	111	138	161	0	0	128	0
379		288	84	94	142	161	0	0	128	0
380		218	85	84	143	158	0	0	128	i
381	d	218	88	88	145	161	0	0	128	0
382		288	92	78	148	182	0	0	128	0
383		287	93	72	140	158	0	0	128	0
384		284	97	87	144	150	0	0	128	0
385		288	188	78	144	159	0	0	128	0
386		134	181	79	143	168	0	0	128	0
387		187	185	79	139	157	0	0	128	0
388		179	185	81	130	157	0	0	128	0
389	z	173	188	89	133	157	0	0	128	0
390		168	185	188	121	158	0	0	128	0
391		158	185	112	112	152	0	0	128	0
392		158	184	124	118	153	0	0	128	g
393		142	181	137	182	153	0	0	128	0
394		130	181	150	99	157	0	0	128	0
395		124	188	170	97	188	0	0	128	58
396		112	98	132	97	187	0	0	128	n
397		90	88	130	98	181	0	0	128	188
398		78	83	133	98	189	0	0	128	188
399		51	79	281	188	179	0	0	128	188
488	$n(d)$	23	78	285	181	172	0	0	128	58
481		16	77	211	184	158	0	0	128	e
482		6	78	288	187	144	0	0	128	0
483	$REL \rightarrow$	1	78	181	188	142	38	48	128	d

404	τ	16	76	164	118	148	20	0	128	0
405		78	70	153	115	158	28	0	128	0
406		148	72	146	121	159	15	0	128	0
407	σ	149	69	142	136	169	10	0	128	0
408		180	67	149	141	178	0	0	128	0
409		29	68	153	143	178	0	0	128	0
410		18	69	158	144	173	0	0	128	0
411		3	71	157	144	178	0	0	128	0
412		0	72	63	144	171	0	0	128	0
413	p	0	74	56	142	173	0	0	128	0
414		0	70	64	142	178	0	0	200	0
415	\rightarrow	0	77	97	174	182	30	40	200	0
416	REL	0	79	187	158	186	25	0	200	0
417		0	82	118	136	189	28	0	200	0
418		0	83	123	128	189	15	0	200	0
419		0	85	181	128	182	18	0	200	0
420		0	88	181	144	172	0	0	200	0
421		0	98	146	155	158	0	0	200	0
422		62	92	152	169	155	0	0	128	0
423	e	161	92	123	162	153	0	0	128	0
424		191	98	189	163	155	0	0	128	0
425		209	92	185	158	152	0	0	128	0
426		218	92	185	145	158	0	0	128	0
427		211	93	185	142	161	0	0	128	0
428		211	94	118	135	168	0	0	128	0
429		209	91	114	138	168	0	0	128	0
430		207	98	118	126	162	0	0	128	0
431	\wedge	202	92	148	114	163	0	0	128	0
432		193	92	163	183	178	0	0	128	0
433		72	92	164	94	178	0	0	128	0
434		38	93	205	88	175	0	0	128	0
435	\varnothing	34	93	203	82	175	0	0	128	0
436		28	93	211	79	173	0	0	128	0
437		18	95	192	77	173	0	0	128	0
438	\rightarrow	55	94	177	76	172	0	50	128	0
439	REL	124	92	178	88	165	0	0	128	0
440		174	91	157	85	163	0	0	128	0
441	l	193	89	147	88	168	0	0	128	0
442		199	88	147	98	159	0	0	128	0
443		208	88	152	93	158	0	0	128	0
444		195	88	167	188	158	0	0	128	0
445		187	85	196	183	157	0	0	128	0
446		168	83	198	185	157	0	0	128	0
447	n	142	88	198	187	161	0	0	128	0
448		112	78	195	188	162	0	0	128	0
449		69	76	198	188	163	0	0	128	0
450		36	75	213	188	162	0	0	128	0
451		11	76	185	187	157	0	0	128	0
452	τ	0	76	152	187	151	0	0	128	0
453	\rightarrow	0	75	139	186	146	30	40	128	0
454	REL	16	73	175	187	153	25	0	128	0
455		71	69	156	113	158	28	0	128	0
456		128	64	149	115	168	15	0	128	0
457	σ	165	58	147	117	165	18	0	128	0
458		159	54	151	121	178	0	0	128	0
459		132	49	154	127	168	0	0	128	0
460		63	43	182	138	172	0	0	128	0
461		32	35	221	132	168	0	0	178	0

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462		28	32	217	125	159	0	0	178	0
463		23	43	223	117	158	0	0	178	0
464	<i>j</i>	21	48	235	116	172	0	10	178	0
465		22	32	178	117	178	0	28	178	0
466		41	33	164	117	172	0	38	168	0
467		71	36	134	122	178	0	38	148	0
468		73	38	134	121	168	29	38	138	0
469	<i>o</i>	78	61	131	117	168	38	48	128	0
470		13	63	73	113	167	38	48	128	0
471		18	63	43	113	169	28	48	128	0
472		13	63	31	180	164	18	48	128	0
473	<i>s</i>	13	64	38	188	167	0	48	128	0
474		8	66	37	188	156	0	38	128	0
475		8	66	41	188	147	0	28	128	0
476		8	67	44	181	153	0	0	128	0
477		4	78	37	186	165	0	0	128	0
478		0	72	39	188	157	0	0	128	0
479		0	73	47	187	133	0	0	128	0
480		0	73	53	188	131	0	0	128	0
481		0	77	67	189	144	0	0	128	0
482	<i>t</i>	0	77	78	187	148	0	0	128	0
483	<i>→</i>	12	78	189	184	143	38	48	128	0
484		34	79	141	186	138	29	0	128	0
485		64	77	133	112	166	0	0	128	0
486		132	73	116	112	169	0	0	128	0
487	<i>æ</i>	171	74	181	111	169	0	0	128	0
488		188	73	98	118	168	0	0	128	0
489		183	74	84	118	163	0	0	128	0
490		183	74	88	111	162	0	0	128	0
491		182	77	98	118	168	0	0	128	0
492		179	88	89	187	162	0	0	128	0
493		176	81	187	188	168	0	0	128	0
494		178	84	121	184	137	0	0	128	0
495		163	84	133	187	136	0	0	128	0
496		134	83	131	189	137	0	0	128	0
497		143	81	288	189	138	0	0	128	0
498	<i>nd</i>	133	81	281	189	168	0	0	128	0
499		189	88	284	189	163	0	0	128	0
500		78	77	287	189	171	0	0	128	0
501		38	74	288	189	179	0	0	128	0
502		38	72	193	189	188	0	0	128	0
503		38	71	162	118	183	0	0	128	0
504		68	72	137	112	194	0	0	128	0
505		123	71	131	113	286	0	0	128	0
506		171	69	148	114	221	0	0	128	0
507	<i>a</i>	166	67	147	113	234	0	0	128	0
508		134	66	131	117	242	0	0	128	0
509		128	61	168	118	242	0	0	128	0
510		91	38	184	111	233	0	0	128	0
511		43	36	211	112	224	0	0	128	0
512		38	33	281	113	213	0	0	128	0
513		37	34	284	111	288	0	0	128	0
514		34	38	282	112	281	0	0	128	0
515		38	47	213	116	194	0	0	128	0
516		28	44	287	119	194	0	0	128	0
517	<i>(d)</i>	24	42	214	126	288	0	0	128	0
518		23	39	214	129	282	0	0	128	0
519		23	33	218	131	287	0	0	128	0

020	20	32	227	141	206	0	0	128	0
021	20	30	224	147	204	0	0	128	0
022	16	30	225	152	202	0	0	128	0
023	15	47	136	151	166	0	0	128	0
024	119	54	117	150	174	0	0	128	0
025	156	57	102	153	170	0	0	128	0
026	195	57	96	154	167	0	0	128	0
027	216	57	89	153	164	0	0	128	0
028	223	57	84	153	161	0	0	128	0
029	226	57	81	153	161	0	0	128	0
030	230	58	79	153	160	0	0	128	0
031	230	59	81	150	162	0	0	128	0
032	215	62	100	147	163	0	0	128	0
033	200	64	92	146	167	0	0	128	0
034	180	64	93	143	173	0	0	128	0
035	167	66	99	138	167	0	0	128	0
036	140	67	109	135	168	0	0	100	0
037	115	67	117	133	166	0	0	100	0
038	79	65	130	130	167	0	0	100	0
039	30	61	137	125	164	20	30	100	0
040	12	56	140	120	167	30	50	100	0
041	2	56	142	113	167	30	50	100	0
042	0	60	147	107	169	30	50	100	0
043	0	60	147	102	169	30	50	100	0
044	0	61	150	101	175	30	50	100	0
045	0	62	151	102	167	30	50	100	0
046	0	62	151	107	158	30	50	100	0
047	0	63	146	116	151	30	50	100	0
048	0	63	137	136	162	20	30	100	0
049	12	64	131	143	167	0	0	100	0
050	48	64	111	143	171	0	0	100	0
051	75	55	99	143	173	0	0	100	0
052	90	48	66	145	176	0	0	128	0
053	100	51	79	149	176	0	0	128	0
054	117	54	72	152	173	0	0	128	0
055	131	54	68	153	165	0	0	128	0
056	136	57	68	150	163	0	0	128	0
057	140	57	72	146	166	0	0	128	0
058	136	58	76	143	165	0	0	128	0
059	135	57	79	139	168	0	0	128	0
060	132	57	91	133	165	0	0	128	0
061	131	57	103	123	164	0	0	128	0
062	128	57	122	114	164	0	0	128	0
063	126	57	146	107	164	0	0	128	0
064	123	57	172	103	166	0	0	128	0
065	121	56	201	100	167	0	0	128	0
066	119	55	211	98	166	0	0	128	0
067	116	54	213	96	164	0	0	128	0
068	119	55	217	94	162	0	0	128	0
069	120	55	217	91	161	0	0	128	0
070	123	55	219	90	161	0	0	128	0
071	125	54	214	89	164	0	0	128	0
072	125	56	209	89	171	0	0	128	0
073	126	54	208	91	181	0	0	128	0
074	129	54	183	95	209	0	0	128	0
075	131	52	160	106	210	0	0	128	0
076	134	54	146	131	211	0	0	128	0
077	136	54	140	150	211	0	0	128	0

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070	143	54	133	166	214	0	0	126	0
073	149	52	124	173	214	0	0	126	0
080	157	52	123	182	214	0	0	126	0
081	161	52	123	188	211	0	0	126	0
082	167	51	121	186	214	0	0	126	0
083	171	51	121	185	222	0	0	126	0
084	172	51	116	185	232	0	0	126	0
085	174	52	112	175	234	0	0	126	0
086	173	51	110	169	255	0	0	126	0
087	172	54	110	167	255	0	0	126	0
088	166	57	109	158	255	0	0	126	0
089	157	57	112	151	255	0	0	126	0
090	151	58	113	140	254	0	0	126	0
091	71	60	112	126	245	0	0	126	0
092	19	62	113	123	238	0	0	126	0
093	11	64	114	113	224	0	0	126	0
094	4	66	116	98	213	0	0	126	0
095	1	67	147	89	204	0	0	126	0
096	0	69	149	86	203	0	0	126	0
097	0	72	146	84	196	0	0	126	0
098	0	74	147	83	193	0	0	126	0
099	0	76	147	81	190	0	0	126	0
098	0	79	142	80	189	0	0	126	0
081	0	81	136	80	186	0	0	126	0
082	2	82	137	80	181	0	0	126	0
083	0	84	141	79	173	0	0	126	0
084	15	86	142	78	168	0	0	126	0
085	46	88	142	78	169	0	0	126	0
086	82	89	147	79	172	0	0	126	0
087	118	92	152	79	172	0	0	126	0
088	154	93	151	76	163	0	0	126	0
089	172	100	150	74	158	0	0	126	0
010	164	101	149	70	152	0	0	126	0
011	193	103	155	68	146	0	0	126	0
012	199	105	166	67	141	0	0	126	0
013	205	108	171	64	141	0	0	126	0
014	206	108	175	66	141	0	0	126	0
015	203	109	176	67	140	0	0	126	0
016	194	108	180	69	141	0	0	126	0
017	120	108	186	73	142	0	0	126	0
018	52	108	186	76	140	0	0	126	0
019	58	105	186	77	139	0	0	126	0
020	96	110	183	76	136	0	0	126	0
021	123	114	181	69	135	0	0	126	0
022	134	113	180	65	133	0	0	126	0
023	137	109	182	60	129	0	0	126	0
024	139	109	183	59	121	0	0	126	0
025	141	109	185	59	120	0	0	126	0
026	139	97	185	62	118	0	0	126	0
027	127	85	186	62	120	0	0	126	0
028	105	68	183	65	123	0	0	126	0
029	62	55	164	70	126	0	0	126	0
030	38	46	157	71	133	0	0	126	0
031	25	46	147	74	138	0	0	126	0
032	30	74	133	78	141	0	0	126	0
033	36	80	126	80	146	0	0	126	0
034	51	80	126	79	152	0	0	126	0
035	62	81	131	80	156	0	0	126	0

626		115	61	151	61	159	0	0	126	0
627		141	61	146	60	164	0	0	126	0
628		172	64	155	60	163	0	0	126	0
630		199	65	159	79	164	0	0	126	0
640	<i>ci</i>	222	69	161	76	159	0	0	126	0
641		234	66	163	75	154	0	0	126	0
642		239	65	165	74	157	0	0	126	0
643		242	64	168	72	155	0	0	126	0
644		237	64	176	74	156	0	0	126	0
645		215	61	172	60	155	0	0	126	0
646		166	61	175	66	159	0	0	126	0
647		58	62	165	99	162	0	0	126	0
648	<i>λ</i>	56	64	163	100	158	0	0	126	0
649		97	64	179	61	151	0	0	126	0
658		161	65	166	75	145	0	0	126	0
651		179	65	191	69	146	0	0	126	0
652		166	66	194	68	159	0	0	126	0
653	<i>i</i>	163	68	196	65	146	0	0	126	0
654		159	68	197	64	142	0	0	126	0
655		123	68	196	66	144	0	0	126	0
656		92	68	194	66	146	0	0	126	0
657		66	66	166	69	151	0	0	126	0
658		56	65	164	75	160	0	0	126	0
659	<i>m</i>	56	77	201	68	167	0	0	126	0
660		66	69	209	116	174	0	0	126	0
661		79	66	219	125	176	0	0	126	0
662		65	63	212	120	176	0	0	126	0
663		96	61	214	116	179	0	0	126	0
664		95	66	216	112	179	0	0	126	0
665		110	60	219	115	177	0	0	126	0
666	<i>a</i>	126	60	219	112	175	0	0	126	0
667		126	61	217	115	175	0	0	126	0
668		144	61	116	145	171	0	0	126	0
669		155	62	79	155	172	0	0	126	0
670		166	63	75	142	176	0	0	126	0
671		162	64	75	156	172	0	0	126	0
672		199	64	72	155	175	0	0	126	0
673	<i>z</i>	216	67	75	124	176	0	0	126	0
674		212	74	91	116	162	0	0	126	0
675		212	76	105	115	166	0	0	126	0
676		266	77	155	166	194	0	0	126	0
677		196	76	146	162	203	0	0	126	0
678		167	76	176	96	200	0	0	126	0
679		47	79	172	100	206	0	0	100	0
680		9	78	167	96	200	0	0	100	0
681		0	76	171	102	166	0	0	100	0
682		0	79	151	116	195	0	0	100	0
683	<i>z</i>	0	77	125	116	196	56	46	100	0
684		0	77	161	122	203	25	0	100	0
685		0	77	162	155	216	27	0	100	0
686		10	76	126	157	216	15	0	100	0
687		56	76	166	166	255	16	0	100	0
688	<i>r</i>	99	77	156	165	245	0	0	100	0
689		169	74	156	165	247	0	0	100	0
690		195	72	154	164	257	0	0	100	0
691		266	71	152	166	256	0	0	100	0
692	<i>o</i>	212	68	155	157	215	0	0	100	0
693		211	67	154	156	196	0	0	100	0

694	223	00	137	171	168	0	0	150	0
695	191	00	140	176	161	0	0	160	0
696	144	00	132	176	174	0	0	160	0
697	91	06	169	179	175	0	0	160	0
698	19	01	169	182	176	0	0	160	0
699	0	01	149	176	176	0	0	160	0
700	0	01	125	174	171	0	0	160	0
701	0	02	91	118	137	0	0	160	0
702	0	00	69	186	146	0	0	160	0
703	0	06	110	111	141	30	40	160	0
704	0	06	146	115	146	20	0	160	0
705	0	06	140	116	176	29	0	160	0
706	0	07	162	138	190	10	0	160	0
707	40	07	173	136	197	10	0	190	0
708	64	07	202	167	194	0	0	160	0
709	70	07	207	161	190	0	0	190	0
710	90	00	206	161	189	0	0	190	0
711	70	04	203	158	192	0	0	160	0
712	00	03	190	149	182	0	0	120	0
713	0	03	198	126	182	0	0	120	0
714	0	03	221	126	181	0	0	120	0
715	0	06	214	114	182	0	0	120	0
716	0	08	126	187	190	0	0	120	0
717	0	01	124	181	161	27	40	120	0
718	0	03	119	96	174	16	0	120	0
719	0	04	190	93	174	19	0	120	0
720	60	04	174	88	174	0	0	120	0
721	116	04	179	86	174	0	0	120	0
722	171	04	177	88	175	0	0	120	0
723	190	07	170	90	168	0	0	120	0
724	207	07	171	107	163	0	0	120	0
725	211	07	170	112	167	0	0	120	0
726	213	07	166	118	168	0	0	120	0
727	210	04	159	123	190	0	0	120	0
728	208	03	152	127	190	0	0	120	0
729	200	00	147	128	190	0	0	120	0
730	198	04	146	128	198	0	0	120	0
731	196	46	148	128	204	0	0	120	0
732	146	41	157	128	212	0	0	120	0
733	120	00	150	127	219	0	0	120	0
734	120	00	164	126	231	0	0	120	0
735	92	00	158	126	240	0	0	120	0
736	0	04	142	126	247	0	0	120	0
737	0	04	147	126	248	0	0	120	0
738	0	04	139	128	250	0	0	120	0
739	0	00	141	128	250	0	0	120	0
740	0	00	142	129	250	0	0	120	0
741	0	00	149	129	250	0	0	120	0
742	0	00	148	131	250	0	0	120	0
743	0	00	147	132	250	0	0	120	0
744	10	00	138	134	250	0	0	120	0
745	11	00	130	139	250	0	0	120	0
746	0	00	134	142	250	0	0	120	0

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Descriptions of COMPUTALKER Speech Parameters
Data Tapes DIGITS and LETTERS

DIGITS data dated July 30, 1977:

Length 400 frames (occupies 3600 bytes).

Contains parameter data for the 10 digits 0,1,2,3,4,5,6,7,8, and 9 in CT-1 Speech Parameter Data format.

digit	begins at frame	
0	1	All digits are 40 frames long
1	41	
2	81	
3	121	
4	161	
5	201	
6	241	
7	281	
8	321	
9	361	

LETTERS data dated Mar 28, 1977:

Length 1040 frames (occupies 9360 bytes).

Contains parameter data for the 26 letters A thru Z in CT-1 Speech Parameter Data format.

letter	begins at frame	letter	begins at frame
A	1	N	521
B	41	O	561
C	81	P	601
D	121	Q	641
E	161	R	681
F	201	S	721
G	241	T	761
H	281	U	801
I	321	V	841
J	361	W	881
K	401	X	921
L	441	Y	961
M	481	Z	1001

All letters are 40 frames long

INTEL HEX FORMAT

The assembled Hex Object paper tape of the CSR1 system is punched in INTEL HEX format. The following is a definition of that format.

Frame 0	Record Mark. Signals the start of a record. The ASCII character colon (":" HEX 3A) is used as the record mark.
Frames 1,2 (0-9,A-F)	Record Length. Two ASCII characters representing a hexadecimal number in the range 0 to 'FF' (0 to 255). This is the count of actual data bytes in the record type or checksum. A record length of 0 indicates end of file.
Frames 3 to 6	Load Address. Four ASCII characters that represent the initial memory location where the data following will be loaded. The first data byte is stored in the location pointed to by the load address; succeeding data bytes are loaded into ascending addresses.
Frames 7, 8	Record Type. Two ASCII characters. Currently all records are type 0. This field is reserved for future expansion.
Frames 9 to 9+2* (Record Length) -1	Data. Each 8 bit memory word is represented by two frames containing the ASCII characters (0 to 9, A to F) to represent a hexadecimal value 0 to 'FF'H (0 to 255).
Frames 9+2* (Record Length) to 9+2* (Record Length) +1	Checksum. The checksum is the negative of the sum of all 8 bit bytes in the record since the record mark (":") evaluated modulus 256. That is, if you add together all the 8 bit bytes, ignoring all carries out of an 8-bit sum, then add the checksum, the result is zero.

Example: If memory locations 1 through 3 contain 53F8EC, the format of the hex file produced when these locations are punched is:

:0300010053F8ECC5

```

0080          0010 * INTEL HEX FORMAT LOADER
0080          0020 *
0080          0030 * COPIED FROM IMSAI SCS REV 1
0080          0040 * TO USE THIS LOADER, FIRST KEY IT IN, STARTING
0080          0050 * AT LOCATION 0080H. THEN MOUNT THE TAPE IN THE
0080          0060 * TTY READER, SET THE ADDRESS SWITCHES TO 0080H,
0080          0070 * THEN PRESS 'STOP', 'RESET', 'EXAMINE', AND 'RUN'.
0080          0080 * THE TTY READER SHOULD START AUTOMATICALLY.
0080          0090 * FOR A DIFFERENT READER INPUT ADDRESS, MODIFY
0080          0100 * ROUTINE 'CHRIN' AT 00E0H.
0080          0110 *
0080          0120 *
0080          0130 TTYS EQU 0
0080          0140 TTYD EQU 1
0080          0150 *
0080 31 80 00 0160 HXLOAD LXI SP,$ * SET UP THE STACK
0083 3E CE 0170 MVI A,0CEH
0085 D3 00 0180 OUT TTYS * SET TTY MODE
0087 3E 17 0190 MVI A,17H
0089 D3 00 0200 OUT TTYS
008B 3E 11 0210 MVI A,11H
008D D3 01 0220 OUT TTYD * X-ON TO READER
008F CD 95 00 0230 CALL HEXIN * READ THE TAPE
0092 C3 92 00 0240 JMP $ * THEN HANG
0095          0250 *
0095          0260 * HEX LOAD SUBROUTINE BEGINS HERE
0095 CD E0 00 0270 HEXIN CALL CHRIN * GET A CHAR
0098 FE 3A 0280 CPI ':'
009A C2 95 00 0290 JNZ HEXIN * WAIT FOR COLON
009D 0E 00 0300 MVI C,0 * CLEAR CHECKSUM
009F CD C2 00 0310 CALL GEBYT * GET THE COUNT
00A2 B7 0320 ORA A * SET THE FLAGS
00A3 C8 0330 RZ * RET ZERO IF EOF RECORD
00A4 47 0340 MOV B,A * ELSE PUT COUNT IN B
00A5 CD C2 00 0350 CALL GEBYT * GET HI BYTE OF ADDRESS
00A8 67 0360 MOV H,A
00A9 CD C2 00 0370 CALL GEBYT * GET LO BYTE OF ADDRESS
00AC 6F 0380 MOV L,A * ADDRESS IS IN HL
00AD CD C2 00 0390 CALL GEBYT * GET & IGNORE RECORD TYPE
00B0          0400 * GO THRU THIS LOOP ONCE FOR EACH DATA BYTE IN A RECORD
00B0 CD C2 00 0410 LOOP CALL GEBYT * GET A DATA BYTE
00B3 77 0420 MOV M,A * STORE IT
00B4 23 0430 INX H * BUMP ADDR
00B5 05 0440 DCR B * DECR COUNT
00B6 C2 B0 00 0450 JNZ LOOP * DO IT AGAIN
00B9 CD C2 00 0460 CALL GEBYT * GET THE CHECKSUM
00BC 79 0470 MOV A,C
00BD B7 0480 ORA A * SET FLAGS, RET IF NON-ZERO
00BE C0 0490 RNZ
00BF C3 95 00 0500 JMP HEXIN * ELSE GO GET NEXT RECORD

```

00C2		0510	*		
00C2		0520	*	THIS ROUTINE READS TWO CHARS FROM THE TAPE	
00C2		0530	*	AND ASSEMBLES THEM INTO A BYTE, WHICH IS	
00C2		0540	*	RETURNED IN THE A REGISTER.	
00C2	CD D3 00	0550	GEBYT	CALL	GEDIG * GET A HEX DIGIT
00C5	87	0560		ADD	A * SHIFT LEFT 4 BITS
00C6	87	0570		ADD	A
00C7	87	0580		ADD	A
00C8	87	0590		ADD	A
00C9	57	0600		MOV	D,A * SAVE IT IN D
00CA	CD D3 00	0610		CALL	GEDIG * GET ANOTHER DIGIT
00CD	B2	0620		ORA	D
00CE	57	0630		MOV	D,A * SAVE BYTE IN D
00CF	81	0640		ADD	C
00D0	4F	0650		MOV	C,A * ACCUMULATE CHECKSUM
00D1	7A	0660		MOV	A,D * GET DATA BYTE BACK
00D2	C9	0670		RET	
00D3		0680	*		
00D3		0690	*	THIS ROUTINE READS A HEX DIGIT FROM THE TAPE.	
00D3		0700	*	NOTE THAT IT DOES NO VALIDITY CHECKING.	
00D3	CD E0 00	0710	GEDIG	CALL	CHRIN * GET A CHAR FROM THE TAPE
00D6	FE 3A	0720		CPI	'9'+1
00D8	DA DD 00	0730		JC	GENUM * SKIP ADI IF NUMERIC
00DB	C6 09	0740		ADI	9
00DD	E6 0F	0750	GENUM	ANI	0FH * MASK OUT UPPER 4 BITS
00DF	C9	0760		RET	
00E0		0770	*		
00E0		0780	*	THIS ROUTINE READS A CHAR FROM THE TAPE READER	
00E0	DB 00	0790	CHRIN	IN	TTYS * GET TTY STATUS
00E2	E6 40	0800		ANI	40H
00E4	CA E0 00	0810		JZ	CHRIN * WAIT TILL READY
00E7	DB 01	0820		IN	TTYD * GET THE CHAR
00E9	E6 7F	0830		ANI	7FH * KILL THE PARITY BIT
00EB	C9	0840		RET	
00EC		0850	*		
00EC		0860	*	END OF HEX LOADER	

```

0100      0010 * CTPLAY
0100      0020 *
0100      0030 *   CT-1 DEMONSTRATION PLAYBACK PROGRAM
0100      0040 *
0100      0050 * READ SPEECH PARAMETER DATA TAPE AND PLAY TO
0100      0060 * CT-1 SYNTHESIZER UNDER CONSOLE SWITCH CONTROL
0100      0070 *
0100      0080 *
0100      0090 * CTPLAY IS A MINIMAL CT-1 CONTROL ROUTINE. IT CONTAINS
0100      0100 * THE BASIC CT-1 SETUP AND PLAYBACK SUBROUTINES WHICH
0100      0110 * MAY BE INCORPORATED INTO OTHER PROGRAMS. THE ROUTINE
0100      0120 * 'CTOUT', AS INCLUDED HERE IS SET UP TO RUN IN RAM ONLY.
0100      0130 * LOCATION CTO+1 IS MODIFIED DURING CT-1 PLAYBACK.
0100      0140 * THE PROGRAMS CTMON AND CTEDIT CONTAIN EXAMPLES WHERE
0100      0150 * THIS OUTPUT INSTR. IS CONSTRUCTED IN RAM SO THE CODE
0100      0160 * ITSELF CAN BE RUN IN ROM.
0100      0170 *
0100      0180 *
0100      0190 * USING CTPLAY:
0100      0200 *   PUT ALL SWITCHES DOWN
0100      0210 *   RAISE SW 0 TO READ A PAPER TAPE INTO THE BUFFER
0100      0220 *   RAISE SW 6 TO REPEAT CT-1 PLAYBACK
0100      0230 *   RAISE SW 7 TO PLAY DATA TO CT-1
0100      0240 *
0100      0250 *
0100      0260 *
0100      0270 * WRITTEN   DEC 20, 1976
0100      0280 * VERSION 1.1   REVISED SEP 22, 77
0100      0290 * BY LLOYD RICE
0100      0300 *   COMPUTALKER CONSULTANTS
0100      0310 *   P.O. BOX 1951
0100      0320 *   SANTA MONICA, CA 90406
0100      0330 *
0100      0340 *
0100      0350 *
0100      0360 * THIS CONTROL MONITOR SOFTWARE IS FURNISHED TO PURCHASERS
0100      0370 * OF THE COMPUTALKER MODEL CT-1 SPEECH SYNTHESIZER.
0100      0380 * IT MAY BE COPIED OR MODIFIED AS DESIRED WITHOUT
0100      0390 * SPECIFIC PERMISSION. COMPUTALKER'S RESPONSIBILITY FOR
0100      0400 * MAINTENANCE APPLIES ONLY TO THE ORIGINAL VERSION OF
0100      0410 * THE CODE (AS LISTED HERE) AND ONLY WHEN IN USE BY THE
0100      0420 * ORIGINAL PURCHASER.
0100      0430 *
0100      0440 * COMPUTALKER OFFERS TO ASSIST WITH OTHER VERSIONS
0100      0450 * WHEREVER REASONABLE AND POSSIBLE.
0100      0460 *

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0100      0470 *
0100      0480 CTBASE EQU 0E0H
0100      0490 *
0100      0500 *
0100      0510 STACK EQU $
0100      0520 *
0100 31 00 01 0530 CTPLAY LXI SP,STACK
0103 3E 55 0540 START MVI A,55H * SET CONSOLE LIGHTS TO 10101010
0105 D3 FF 0550      OUT 0FFH
0107 01 00 10 0560 SDTIME LXI B,1000H * ALL SWITCHES MUST BE DOWN
010A DB FF 0570 SWDN IN 0FFH * FOR 100 MSEC CONTINUOUSLY
010C B7 0580      ORA A
010D C2 07 01 0590      JNZ SDTIME
0110 0B 0600      DCX B
0111 78 0610      MOV A,B
0112 B1 0620      ORA C
0113 C2 0A 01 0630      JNZ SWDN
0116 DB FF 0640 SWUP IN 0FFH * LOOP HERE UNTIL A SW IS UP
0118 B7 0650      ORA A
0119 CA 16 01 0660      JZ SWUP
011C 17 0670      RAL
011D DA 3F 01 0680      JC PLAY * SW 7 UP, PLAYBACK
0120 17 0690      RAL
0121 DA 47 01 0700      JC REPEAT * SW 6 UP, REPEAT
0124 17 0710      RAL
0125 DA 03 01 0720      JC START * SW 5 UP,
0128 17 0730      RAL
0129 DA 03 01 0740      JC START * SW 4 UP,
012C 17 0750      RAL
012D DA 03 01 0760      JC START * SW 3 UP,
0130 17 0770      RAL
0131 DA 03 01 0780      JC START * SW 2 UP,
0134 17 0790      RAL
0135 DA 03 01 0800      JC START * SW 1 UP,
0138 17 0810      RAL
0139 DA 95 01 0820      JC READ * SW 0 UP, READ PAPER TAPE
013C C3 07 01 0830      JMP SDTIME * HOW DID WE GET HERE??
013F      0840 *
013F      0850 *
013F      0860 * CT-1 PLAYBACK ROUTINES
013F      0870 *
013F      0880 * THE CT-1 PLAYBACK CODE ASSUMES A FRAMECOUNT IS STORED
013F      0890 * IN THE 2 BYTES JUST PRECEEDING THE DATA BUFFER.
013F      0900 *
013F      0910 *
013F      0920 * PLAYBACK
013F      0930 *
013F 3E 7F 0940 PLAY MVI A,7FH * PUT 80 IN CONSOLE LEDS
0141 D3 FF 0950      OUT 0FFH

```

```

0143 AF          0960          XRA   A
0144 C3 4D 01   0970          JMP   PLABK      * CLEAR REPEAT SW
0147           0980 *
0147           0990 * REPEAT
0147           1000 *
0147 3E BF      1010 REPEAT MVI   A,0BFH  * PUT 40 IN CONSOLE LEDS
0149 D3 FF      1020          OUT   0FFH
014B 3E 40      1030          MVI   A,40H
014D           1040 *
014D           1050 * COMMON PLAYBACK LOOP
014D           1060 *
014D 32 6E 01   1070 PLABK  STA   RPTSW+1  * REPEAT SW = 40H TO REPEAT, ELSE 0
0150 21 E9 01   1080          LXI   H,BUFFER * GET PLAYBACK ARGUMENTS
0153 5E          1090          MOV   E,M
0154 23          1100          INX   H
0155 56          1110          MOV   D,M      * FRAME COUNT TO DE
0156 23          1120          INX   H      * DATA ADDRESS TO HL
0157 CD 77 01   1130          CALL  CTOUT    * SET CT-1 PARAMS FROM FRAME 1
015A 1B          1140          DCX   D      * COUNT THAT FRAME
015B 3E FF      1150          MVI   A,255   * TURN ON THE CT-1
015D D3 EF      1160          OUT   CTBASE+15
015F CD 77 01   1170 PLOOP  CALL  CTOUT    * PLAY A FRAME
0162 CD 89 01   1180          CALL  DLY10   * DELAY 10 MSEC
0165 1B          1190          DCX   D
0166 7A          1200          MOV   A,D
0167 B3          1210          ORA   E
0168 C2 5F 01   1220          JNZ   PLOOP   * DONE?
016B DB FF      1230          IN   0FFH    * TEST REPEAT SW
016D E6 00      1240 RPTSW  ANI   0      * SEE IF SW 6 IS STILL UP
016F C2 4D 01   1250          JNZ   PLABK   * YES, PLAY AGAIN
0172 D3 EF      1260          OUT   CTBASE+15 * NO, TURN OFF CT-1
0174 C3 03 01   1270          JMP   START   * AND GO BACK TO CMMD LOOP
0177           1280 *
0177           1290 * CTOUT PLAYS 1 DATA FRAME FROM THE BUFFER.
0177           1300 * ON ENTRY: HL POINTS TO AV OF THE FRAME TO PLAY
0177           1310 * CONTENTS OF DE NOT CHANGED BY THIS SUBR
0177           1320 *
0177 06 E0      1330 CTOUT  MVI   B,CTBASE * RE-INITIALIZE OUTPUT ADDR
0179 0E 09      1340          MVI   C,9
017B 78          1350 CTLP   MOV   A,B
017C 32 81 01   1360          STA   CTO+1
017F 7E          1370          MOV   A,M
0180 D3 E0      1380 CTO    OUT   CTBASE  * OUTPUT THE PARAMETER
0182 23          1390          INX   H
0183 04          1400          INR   B
0184 0D          1410          DCR   C
0185 C2 7B 01   1420          JNZ   CTLP    * GO AROUND 9 TIMES
0188 C9          1430          RET
0189           1440 *

```


0189		1450	*	DELAY 10 MILLISECONDS (ASSUMES 2 MHZ CLOCK)	
0189		1460	*	(A) CHANGED, ALL ELSE RESTORED	
0189		1470	*		
0189	E5	1480	DLY10	PUSH	H
018A	21 20 03	1490		LXI	H,800
018D	2B	1500		DCX	H
018E	7C	1510		MOV	A,H
018F	B5	1520		ORA	L
0190	C2 8D 01	1530		JNZ	S-3
0193	E1	1540		POP	H
0194	C9	1550		RET	
0195		1560	*		
0195		1570	*		
0195		1580	*	PAPER TAPE READER LOOP	
0195		1590	*		
0195	3E FE	1600	READ	MVI	A,0FEH * PUT 01 IN CONSOLE PORT
0197	D3 FF	1610		OUT	0FFH
0199	CD CA 01	1620		CALL	PTRCLR * CLEAR READER FLAG
019C	CD D3 01	1630	PT1	CALL	PTRIN * IGNORE THE 1ST BYTE
019F	CA 9C 01	1640		JZ	PT1 * IF IT TAKES FOREVER
01A2	CD D3 01	1650	PT2	CALL	PTRIN * IGNORE NULLS (LEADER)
01A5	CA 03 01	1660		JZ	START * ZERO IF TIMER RAN OUT
01A8	B7	1670		ORA	A
01A9	CA A2 01	1680		JZ	PT2
01AC	21 E9 01	1690		LXI	H,BUFFER * ALSO IGNORE 1ST NON-ZERO BYTE
01AF	CD D3 01	1700		CALL	PTRIN * READ 2 FRAME COUNT BYTES
01B2	CA 03 01	1710		JZ	START
01B5	77	1720		MOV	M,A
01B6	23	1730		INX	H
01B7	CD D3 01	1740		CALL	PTRIN
01BA	CA 03 01	1750		JZ	START
01BD	77	1760		MOV	M,A
01BE	23	1770		INX	H
01BF	CD D3 01	1780	PT3	CALL	PTRIN * THEN READ IN THE DATA
01C2	CA 03 01	1790		JZ	START
01C5	77	1800		MOV	M,A
01C6	23	1810		INX	H
01C7	C3 BF 01	1820		JMP	PT3 * READ UNTIL TIME-OUT
01CA		1830	*		
01CA		1840	*		
01CA		1850	*	PAPER TAPE I/O	
01CA		1860	*		
01CA		1870	PTRF	EQU	16H * PTR FLAGS PORT
01CA		1880	PTRD	EQU	17H * PTR DATA PORT
01CA		1890	*		

```

01CA 26 03      1900 PTRCLR MVI H,3
01CC D3 16      1910          OUT PTRF
01CE 26 11      1920          MVI H,11H
01D0 D3 16      1930          OUT PTRF
01D2 C9         1940          RET
01D3           1950 *
01D3           1960 * WAIT FOR PTR FLAG & READ 1 BYTE. IF FLAG IS NOT
01D3           1970 * READY WITHIN 400 MSEC, RETURN WITH Z FLAG SET.
01D3           1980 * IF DATA READ NORMALLY, RETURN WITH Z=0, &
01D3           1990 * THE DATA BYTE IS IN A.
01D3           2000 *
01D3 C5         2010 PTRIN  PUSH B          * SAVE B
01D4 01 00 40   2020          LXI B,4000H      * SET TIMER
01D7 DB 16      2030 P1    IN PTRF          * WAIT FOR FLAG
01D9 00         2040          NOP          * CMA FOR ACTIVE LOW STATUS
01DA E6 01      2050          ANI 1
01DC C2 E5 01   2060          JNZ P2
01DF 0B         2070          DCX B          * NOT READY, CHECK THE TIME
01E0 78         2080          MOV A,B
01E1 B1         2090          ORA C
01E2 C2 D7 01   2100          JNZ P1
01E5 DB 17      2110 P2    IN PTRD          * CONDITION ZERO IF TIME OUT
01E7 C1         2120          POP B          * RESTORE B
01E8 C9         2130          RET
01E9           2140 *
01E9           2150 *
01E9           2160 * CT-1 PARAMETER BUFFER
01E9           2170 *
01E9 00         2180 BUFFER DB 0          * CT-1 DATA BUFFER BEGINS HERE
01EA           2190 *          * 1ST 2 BYTES OF BUFFER ARE FRAME COUNT
01EA           2200 *          * THEN FOLLOWS THE PARAMETER DATA
01EA           2210 *

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SYMB

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CTBASE  00E0 : STACK      0100 : CTPLAY      0100
START   0103 : SDTIME     0107 : SWDN       010A
SWUP    0116 : PLAY       013F : REPEAT     0147
PLABK   014D : PLOOP      015F : RPTSW      016D
CTOUT   0177 : CTLP       017B : CTO        0180
DLY10   0189 : READ       0195 : PT1         019C
PT2     01A2 : PT3        01BF : PTRF         0016
PTRD    0017 : PTRCLR     01CA : PTRIN       01D3
P1      01D7 : P2         01E5 : BUFFER       01E9

```

OPCD OPERAND

* LINE # IN E. CHAR POS IN D
 * DISPLAY PARTIAL LINE 1 HEADER
 * DISPLAY COMMAND PROMPT
 * INIT PLAY PARAMETERS
 * FRAME COUNT (= LENGTH
 * BEGIN AT FRAME 1
 * INIT CURRENT LINE TO LINE 1
 * INIT CURRENT PARAM TO RV

OPCD OPERAND

CT-1 CONTROL MONITOR
 VERSION 1.10
 ALLOWS READING, EDITING, AND STORING CT-1
 SPEECH PARAMETER DATA FILES IN DECIMAL USING
 A POLYMORPHIC OR PROFESSOR TECH VIDEO DISPLAY.
 SUPPORTS STORAGE ON PAPER TAPE OR AUDIO
 CASSETTE IN THE TRUBELL FORMAT.
 WRITTEN DEC 18, 1976
 BY LLOYD RICE
 COMPUTALKER CONSULTANTS
 P. O. BOX 1951
 SANTA MONICA, CA 90406

THIS CONTROL MONITOR SOFTWARE IS FURNISHED TO
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 WHEREVER REASONABLE AND POSSIBLE.

ORG 100H
 HIGH ORDER VIDEO HOUR
 CT-1 CHAN RV ADDRESS
 INIT LENGTH TO 1 FRAME
 ZERO THE 1ST FRAME
 SET UP PLAUDT SUBROUTINE
 SET UP RETURN
 INITIALIZE THE STACK
 INIT FRAME TIME TO 10 MSEC
 INIT REPEAT DELAY TO 800 MSEC
 CLEAR THE SCREEN
 DISPLAY LINE 0 HEADER

OPCD OPERAND

* LINE # IN E. CHAR POS IN D
 * DISPLAY PARTIAL LINE 1 HEADER
 * DISPLAY COMMAND PROMPT
 * INIT PLAY PARAMETERS
 * FRAME COUNT (= LENGTH
 * BEGIN AT FRAME 1
 * INIT CURRENT LINE TO LINE 1
 * INIT CURRENT PARAM TO RV

* FORCE CL TO WITHIN LEGAL LIMITS
 * LINE #1, CHAR POS 20
 * DISPLAY REST OF HEADER
 * DEFENDING ON CURRENT PARAMETER
 * CPCS, DISPLAY FORWARDTS
 * CT-4, DISPLAY AMPLITUDES
 * TOP DISPLAY LINE IS FRAME CL-4
 * PUT THAT FRAME NO. IN BC
 * BEGIN DISPLAY AT LINE 3
 * DISPLAY LINE COUNTER
 * SAVE IN A TEMP
 * FRAME OUT OF RANGE, CLEAR LINE
 * DISPLAY 1 FRAME OF DATA
 * DISPLAY A BLANK LINE
 * INCREMENT FRAME NO.
 * INCR. DISPLAY LINE NO.
 * DECREMENT LINE COUNT
 * LOOP TO DISPLAY ALL LINES
 * PLACE MARK BY CURENT PARAM.
 * HORIZONTAL POSITION DEPENDS

ADDR	B1	B2	B3	E	LINE	LABEL	OPCD	OPERAND	PCHL	OPCD	OPERAND
018E	24	45	00		0560		LHLD	NUMBER			
01C1	22	43	00		0565		SHLD	LAST			
01C4	1E	0F			0570	CT3H	MVI	E, BPH			
01C6	16	1D			0575		MVI	D, 1DH			
01C8	CD	E1	06		0580		CALL	VBUFFD			
01CB	EB				0585		XCHG				
01CC	22	51	00		0590		SHLD	CMADDR			
01CF	06	0A			0595		MVI	B, 0AH			
01D1	36	00			0600	CT3B	MVI	M, '+128			
01D3	23				0605		INX	H			
01D4	05				0610		DCR	B			
01D5	C2	D1	01		0615		JNZ	CT3B			
01D8	21	00	00		0620	CT4	LXI	H, 0			
01DA	22	45	00		0625		SHLD	NUMBER			
01DE	AF				0630		XRA	A			
01DF	32	47	00		0635	CT4H	STA	BRANCH			
01E2	CD	6E	06		0640	CT5	CALL	GETCMC			
01E5	FE	30			0645		CPI	'0'			
01E7	DA	0F	02		0650		JC	CT5H			
01EA	FE	3A			0655		CPI	'9'+1			
01EC	D2	0F	02		0660		JNC	CT5H			
01EF	E6	0F			0665		ANI	0FH			
01F1	4F				0670		MOV	C, H			
01F2	06	00			0675		MVI	B, 0			
01F4	24	45	00		0680		LHLD	NUMBER			
01F7	29				0685		DAD	H			
01F8	54				0690		MOV	D, H			
01F9	5D				0695		MOV	E, L			
01FA	29				0700		DAD	H			
01FB	29				0705		DAD	H			
01FC	19				0710		DAD	H			
01FD	09				0715		DAD	B			
01FE	24	45	00		0720		SHLD	NUMBER			
0201	3A	47	00		0725		LDA	BRANCH			
0204	07				0730		ORA	A			
0205	C2	E2	01		0735		JNZ	CT5			
0208	3C				0740		INR	H			
0209	32	47	00		0745		STA	BRANCH			
020C	C3	E2	01		0750		JMP	CT5			
020F	FE	00			0755	CT5H	CPI	0DH			
0211	C4	28	02		0760		JNZ	CT5C			
0214	3A	47	00		0765		LDA	BRANCH			
0217	07				0770		ORA	A			
0218	3C				0775		JNZ	CT5B			
021B	3C				0780		INR	H			
021C	07				0785	CT5B	ADD	H			
021D	4F				0790		MOV	C, H			
021E	06	00			0795		MVI	B, 0			
0220	21	5H	07		0800		LXI	H, BRNCTR-2			
0223	09				0805		DAD	B			
0224	E6				0810		XCHG				
0225	1H				0815		LDAX	D			
0226	6F				0820		MOV	L, A			
0227	13				0825		INX	D			
0228	1H				0830		LDAX	D			
0229	67				0835		MOV	H, H			

* ON PARAMETERS DISPLAYED.
 * SET LAST=NUMBER
 * GET CMD LINE VIDEO BUFF ADDR
 * SAVE IT
 * AND CLEAR THE COMMAND LINE
 * SET NUMBER=0
 * SET BRANCH=0
 * GET H CHAR FROM KEYBOARD
 * IT'S C ASCII '0'
 * IT'S > ASCII '9'
 * KEEP THE DECIMAL DIGIT
 * MULTIPLY NUMBER BY 10
 * AND ADD DIGIT
 * IS CHAR A LR ?
 * YES: DO BRANCH OPERATION
 * IF NO BRANCH SET, MAKE IT A 1
 * BUILD POINTER TO BRANCH TABLE

* DO BRANCH
 * SAVE CHAR IN B
 * BRANCH NOT 0, BAD
 * GET THE CHAR BACK
 * BRANCH LIST COUNTER
 * SEARCH BRANCH CHAR TABLE
 * WAS NOT H BRANCH COMMAND
 * SET BRANCH TO CORRESP. VALUE
 * SET POINTERS FOR 'A-' PARAMETER
 * APTAB COUNTER
 * SET POINTERS FOR 'F-' PARAMETER
 * FPTAB COUNTER
 * WAS NOT H GOOD PARAMETER NAME
 * SET CPM TO NEW PARAMETER
 * WHEN CHANGING PARAMETERS,
 * KILL THE ARROW
 * CASSETTE
 * TAPE OR ITY

ADDR	B1	B2	B3	E	LINE	LABEL	OPCD	OPERAND	COMMENT
0207	04	02	02		1120	KILCHK	JZ	KILCHK	
0208	01	04	00		1125	BADCMD	LXI	SP, STACK	* RESET STACK
0209	01	05	00		1130		LXI	H, ILLEG	
0210	01	09	00		1135		LXI	B, 0	
0211	01	0F	00		1140		LXI	D, ID6FH	
0212	0D	F7	06		1145		CALL	DSTX	* DISPLAY "ILLEGAL !"
0213	0D	5B	07		1150	CTSM	CALL	KEYTST	* UNTIL H KEY IS HIT
0214	0A	09	02		1155		JZ	CTSM	
0215	05	04	01		1160		JMP	CT3H	
0216	0D	6E	06		1165	KILCHK	CALL	GETCMC	* TO KILL REQUIRES NAME "KILL"
0217	0E	49			1170		CPI	'I'	
0218	02	02	02		1175	BADCMD	JNZ	BADCMD	
0219	0D	6E	06		1180		CALL	GETCMC	
0220	FE	4C			1185		CPI	'L'	
0221	02	02	02		1190	BADCMD	JNZ	BADCMD	
0222	0D	6E	06		1195		CALL	GETCMC	
0223	FE	4C			1200		CPI	'L'	
0224	02	02	02		1205	BADCMD	JNZ	BADCMD	
0225	03	00	01		1210		JMP	CTMON	
0226					1215				* COMMAND PROCESSORS:
0227	0D	6E	06		1220	CHS10	CALL	GETCMC	
0228	FE	41			1225		CPI	'H'	
0229	0A	F2	02		1230	CHSHP	JZ	CHSHP	
0230	FE	52			1245		CPI	'R'	
0231	0A	0B	03		1250		JZ	CHSRD	
0232	FE	57			1255		CPI	'M'	
0233	0A	0B	03		1260		JZ	CHSMR	
0234	03	0A	02		1265	BADCMD	JMP	BADCMD	
0235	0A	0B	03		1270	CHSHP	CALL	SETAP	* SETUP TO APPEND DATA
0236	0D	H5	03		1275		CALL	CRDATA	* READ IT
0237	2A	56	00		1280	CHS1	LHLD	PLACNT	
0238	EB				1285		XCHG		
0239	2A	4C	00		1290		LHLD	LENGTH	* LENGTH=LENGTH+
0240	19				1295		DAD	D	* NO. OF FRAMES READ
0241	22	4C	00		1300		SHLD	LENGTH	* AFTER READING IN NEW DATA,
0242	3E	20			1305	KARR0W	CALL	AR0W	* KILL THE ARROW
0243	03	00	01		1310		CALL	AR0W	
0244	03	00	01		1315		JMP	CTB	
0245	21	4E	00		1320	CHSRD	LXI	H, BUFFER	* READ DATA
0246	0D	H5	03		1325		CALL	CRDATA	* LENGTH=NO. OF FRAMES READ
0247	2A	56	00		1330	CHS2	LHLD	PLACNT	* LENGTH=NO. OF FRAMES READ
0248	22	4C	00		1335		SHLD	LENGTH	* SETUP BYTE COUNT IN DF
0249	03	00	01		1340	CHSMR	JMP	KARR0W	* INCLUDE FRAME COUNT BYTES
0250	0D	6A	03		1345		CALL	SETAP	
0251	13				1350		INX	D	* WRITE IT
0252	13				1355		INX	D	* DATA IN BUFFER IS STILL OK
0253	0D	04	01		1360		CALL	CRDATA	
0254	03	00	01		1365		JMP	CT3H	
0255					1370				* TAPE TAPE MNU. TTY HANDLERS
0256	0D	6E	06		1375				
0257	FE	41			1380	TAP10	CALL	GETCMC	
0258	0A	0B	03		1385		CPI	'H'	
0259	0A	0B	03		1390		JZ	TAPAP	

ADDR B1 B2 B3 C LINE LABEL UP/D OVER/ND UP/D OVER/ND

ADDR	B1	B2	B3	C	LINE	LABEL	UP/D	OVER/ND	UP/D	OVER/ND
0376	LD	08	07		1688	TR3	CALL	PTRIN		* THEN READ THE REST
037E	LD	08	07		1689	R2	MOV	M, H		* INTO THE BUFFER
0384	LD	08	07		1690		INX	H		
0386	LD	08	07		1695		JMP	TR3		
0388	LD	08	07		1700					
038A	LD	08	07		1701					* PAPER TAPE PUNCH ROUTINE (PRESENTLY UNIMPLEMENTED)
038C	LD	08	07		1702					
038E	LD	08	07		1703					
0390	LD	08	07		1705	IFDATA	RET			
0392	LD	08	07		1710					
0394	LD	08	07		1715					
0396	LD	08	07		1720					
0398	LD	08	07		1725	CRDATA	PUSH	H		* SAVE DATA START ADDR
039A	LD	08	07		1730		MOV	M, H		
039C	LD	08	07		1735		CALL	CASCLR		* INITIALIZE CASSETTE READER
039E	LD	08	07		1740		LXI	M, PLACNT		
03A0	LD	08	07		1745		CALL	CASIN		* READ FRAME COUNT INTO PLACNT
03A2	LD	08	07		1750		CALL	CASIN		
03A4	LD	08	07		1755		LHLD	PLACNT		
03A6	LD	08	07		1760		CALL	SM2		
03A8	LD	08	07		1765		POP	H		* MULTIPLY BY 9 & INTO DE
03AA	LD	08	07		1770	CR1	CALL	CASIN		
03AC	LD	08	07		1775		DCX	D		* READ THE DATA
03AE	LD	08	07		1780		XRA	H		
03B0	LD	08	07		1785		ORH	D		
03B2	LD	08	07		1790		ORH	E		
03B4	LD	08	07		1795		JNZ	CR1		
03B6	LD	08	07		1800		RET			
03B8	LD	08	07		1805	CRDATA	CALL	CASCLR		* WRITE SOME LEADER
03BA	LD	08	07		1810		LXI	M, BUFFER-2		
03BC	LD	08	07		1815		MOV	M, B		
03BE	LD	08	07		1820	CR1	MOV	M, B		
03C0	LD	08	07		1825		CALL	CASOUT		* WRITE THE DATA
03C2	LD	08	07		1830		JMX	H		
03C4	LD	08	07		1835		DCX	D		
03C6	LD	08	07		1840		MOV	M, D		
03C8	LD	08	07		1845		ORH	E		
03CA	LD	08	07		1850		JNZ	CR1		
03CC	LD	08	07		1855		MOV	M, B		
03CE	LD	08	07		1860		CALL	CASOUT		* WRITE SOME TRAILER
03D0	LD	08	07		1865		CALL	DLY25		
03D2	LD	08	07		1870		XRA	H		
03D4	LD	08	07		1875		CALL	CASCLR		
03D6	LD	08	07		1880		RET			
03D8	LD	08	07		1885					
03DA	LD	08	07		1890					
03DC	LD	08	07		1895					
03DE	LD	08	07		1900					
03E0	LD	08	07		1905					
03E2	LD	08	07		1910					
03E4	LD	08	07		1915					
03E6	LD	08	07		1920					
03E8	LD	08	07		1925					
03EA	LD	08	07		1930					
03EC	LD	08	07		1935					
03EE	LD	08	07		1940					
03F0	LD	08	07		1945					

* SET FRAME COUNTER
 * ANY KEYS SEEN HIT?
 * NO
 * YES: KILL ANY REPEATS
 * DELAY (RTIME) MSEC
 * PLAY 1 DATA FRAME
 * WAIT FOR LAST FRAME
 * TURN OFF CT-1
 * NO REPEAT, GO BACK
 * END OF WAIT, PLAY AGAIN
 * KEY "INTERRUPT", GO BACK
 * PARAMETER MODIFY, DITTO & NEXT LINE PROCESSORS
 * SET NUM. BACK TO PREV VALUE
 * COMPUTE ADDR OF CURRENT FRAM
 * HND CURRENT PARAM
 * HND SET DATA = NUMBER
 * REDISPLAY CURRENT FRAME
 * SEE IF WE CAN BUMP CL
 * TOO LOW WE CAN'T GO FORWARD
 * IT'S OK, SET CL=CL+1

ADDR	BI	B2	B3	E	LINE	LABEL	OPCD	OPERAND	ADDR	BI	B2	B3	E	LINE	LABEL	OPCD	OPERAND
0062	3D				3945	DCR	H		006E	D				4200	DCR	D	
0063	02	62	06		3950	JNZ	\$-1		006F	02	62	06		4200	JNZ	CLR	
0066	2B				3955	DCX	H		0070	02	62	06		4200	JNZ	CLR	
0067	B4				3960	UKH	H		0071	02	62	06		4200	JNZ	CLR	
0068	B5				3965	UKR	L		0072	02	62	06		4200	JNZ	CLR	
0069	02	5A	06		3970	JR	CLY		0073	02	62	06		4200	JNZ	CLR	
006A	F1				3975	POP	PSW		0074	02	62	06		4200	JNZ	CLR	
006B	09				3980	RET			0075	02	62	06		4200	JNZ	CLR	
006E					3985	*			0076	F5				4200	JNZ	CLR	
006E					3990	*	GET H COMMAND CHAR FROM KBD INTO H		0077	F5				4200	JNZ	CLR	
006E					3995	*	HND DISPLAY IT IN THE COMMAND LINE		0078	F6	00			4200	JNZ	CLR	
006E					4000	*			0079	F7	00			4200	JNZ	CLR	
006E					4005	GETLINE	PUSH H		007A	F7	00			4200	JNZ	CLR	
006F	0D	4F	07		4010	CALL	GETC		007B	F7	00			4200	JNZ	CLR	
0072	FE	1B			4015	CP1	1BH		007C	0A	8A	06		4200	JNZ	CLR	
0074	0A	8A	06		4020	J2	GC1		007D	FE	7F			4200	JNZ	CLR	
0077	FE	7F			4025	CP1	7FH		007E	0A	8A	06		4200	JNZ	CLR	
0079	0A	8A	06		4030	J2	GC1		007F	0A	8A	06		4200	JNZ	CLR	
007C	0A	51	00		4035	LALD	CMDADR		0080	0A	8A	06		4200	JNZ	CLR	
007F	F5				4040	PUSH	PSW		0081	0A	8A	06		4200	JNZ	CLR	
0080	F6	00			4045	ORI	128		0082	0A	8A	06		4200	JNZ	CLR	
0082	77				4050	MOV	M,A		0083	F1				4200	JNZ	CLR	
0083	F1				4055	POP	PSW		0084	23				4200	JNZ	CLR	
0084	23				4060	INX	H		0085	22	51	00		4200	JNZ	CLR	
0085	22	51	00		4065	SHLD	CMDADR		0086	0A	8A	06		4200	JNZ	CLR	
0088	0A	8A	06		4070	POP	H		0087	0A	8A	06		4200	JNZ	CLR	
0089	0A				4075	RET			0088	31	FF	00		4200	JNZ	CLR	
008A	31	FF	00		4080	LXI	SP,STACK		0089	0A	8A	06		4200	JNZ	CLR	
008D	03	04	01		4085	JMP	CT3H		008A	31	FF	00		4200	JNZ	CLR	
0090					4090	*	VIDEO DISPLAY SUBROUTINES:		008B	03	04	01		4200	JNZ	CLR	
0090					4095	*	CLEAR THE SCREEN		008C	03	04	01		4200	JNZ	CLR	
0090					4096	*			008D	03	04	01		4200	JNZ	CLR	
0090					4097	*			008E	03	04	01		4200	JNZ	CLR	
0091	F5				4100	CLRVID	PUSH PSW		008F	03	04	01		4200	JNZ	CLR	
0092	21	00	80		4105	LXI	H,256*VIDBUF		0090	03	04	01		4200	JNZ	CLR	
0093	36	00			4110	CP1	M, *+128		0091	F5				4200	JNZ	CLR	
0097	23				4115	CLR			0092	21	00	80		4200	JNZ	CLR	
0098	2C				4120	INX	H		0093	36	00			4200	JNZ	CLR	
0099	2C				4125	MOV	M,H		0094	2C				4200	JNZ	CLR	
0099	2C				4130	CP1	VIDBUF+4		0095	36	00			4200	JNZ	CLR	
009B	02	95	06		4135	JNZ	CLR		0096	2C				4200	JNZ	CLR	
009E	E1				4140	FUP	H		0097	23				4200	JNZ	CLR	
009F	F1				4145	POP	PSW		0098	2C				4200	JNZ	CLR	
009B	02				4150	RET			0099	2C				4200	JNZ	CLR	
009B	02				4155	*	CLEAR ONE LINE ACROSS THE SCREEN		009A	2C				4200	JNZ	CLR	
009B	02				4156	*			009B	02				4200	JNZ	CLR	
009B	02				4160	CLRVID	PUSH H		009C	02				4200	JNZ	CLR	
009B	02				4165	PUSH	D		009D	02				4200	JNZ	CLR	
009B	02				4170	AVI	D,0		009E	E1				4200	JNZ	CLR	
009B	02				4175	CALL	VBUFFHD		009F	F1				4200	JNZ	CLR	
009B	02				4180	XCHG			009B	02				4200	JNZ	CLR	
009B	02				4185	AVI	D,+08H		009B	02				4200	JNZ	CLR	
009B	02				4190	AVI	M, *+128		009B	02				4200	JNZ	CLR	
009B	02				4195	INX	H		009B	02				4200	JNZ	CLR	

* MOVE 1 LINE UP OR DOWN (E) SCREEN LINES (TOP DOWN)

* MOVE 1 HORIZONTAL LINE UP OR DOWN 1 LINE ON SCREEN

* MOVE H LINE UP

* MOVE H LINE DOWN

* DESTINATION LINE & AN L

* SOURCE COORDINATES IN DE

* CONVERT TO VIDEO BUFF. ADDR

* DESTINATION COORDINATES IN DE

* CONVERT VIDEO LOOKUP LINE NO. & CHAR PEG...

* GET CHAR POS

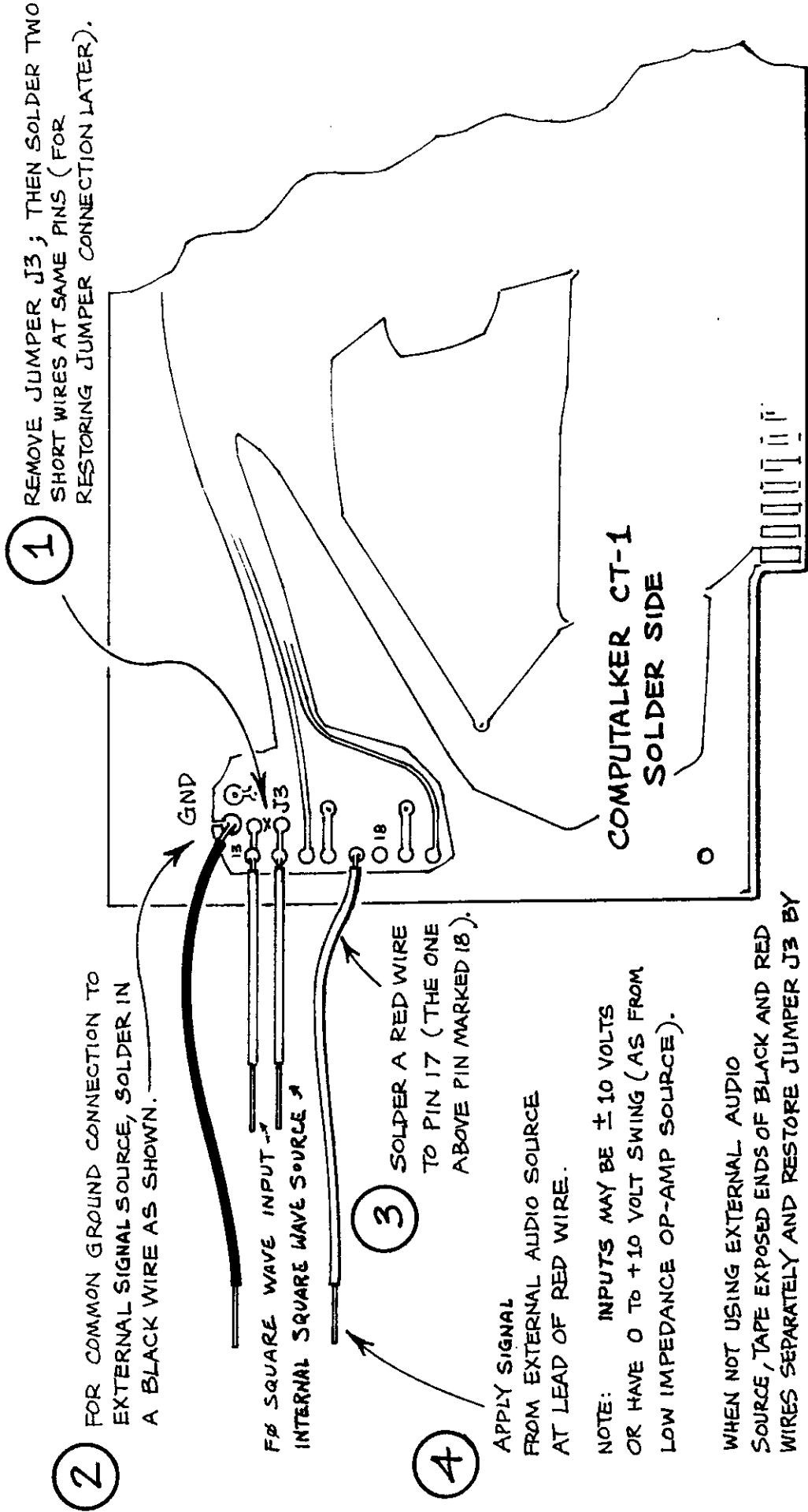
* SAVE CHAR POS IN D

* GET LINE NO.

* SHIFT + LEFT

* PUT VIDEO BUFFER IN VIDEO...

TO ENTER AN AUDIO SQUARE WAVE FROM EXTERNAL SOURCE,
REPLACING AV AND FØ PARAMETERS, FOLLOW PROCEDURE BELOW



Model CT-1 Warranty

The COMPUTALKER Model CT-1 Speech Synthesizer is warranted by Computalker Consultants against defects in workmanship and materials for a period of six (6) months from the date of delivery.

During the warranty period, Computalker Consultants will repair, or at its option, replace at no charge components that prove to be defective provided that the board is returned, shipping prepaid, to:

(if by U.S. Postal Serv.)

COMPUTALKER CONSULTANTS
P.O. Box 1951
Santa Monica, CA 90406

(if by private delivery serv.)

COMPUTALKER CONSULTANTS
1730 A 21st Street
Santa Monica, CA 90404

This warranty does not apply if the board has been damaged by accident or misuse, or as a result of repairs or modifications made by other than authorized personnel at the above captioned service facility.

No other warranty is expressed or implied. Computalker Consultants is not liable for consequential damages.

Because of the critical tuning required for a number of the analog circuits used in the Model CT-1, Computalker Consultants strongly recommends that the board be returned to us for any repairs needed. Beyond the period of the warranty, such repairs will be made with a charge for parts and labor. Computalker Consultants will cover the shipping costs to return the board.

