EMSA
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1. Address clock
2. Address clock + Touch Sensitive KB.
3. Touch Sensitive KB.
4. Touch Sensitive KB.
5. Touch Sensitive KB.
6. D Type Short Term Memory.
7. D Type Short Term Memory.
8. Random Touch Pad and Clock + Real Time DAC.
9. Record And Play Touch Dads + Dynamic Function + Envelope Trigger
10. Record And Play Touch Pads + Memory Control Logic.
11. Shift Register Memory + Gates.
12. Shift Register Memory + Clock Generator - Skipshift.
13. Meter DAC + Clock Generqtor - Skip And Shift.
14. Memory DAC + Transposition Pads + Shift And Skip.
15. Transposition Pads.
16. Transposition Pads + Appendix.

Back Page. Shift Register. MF50052T.
or MF7104.

The pulses generated are used to compile various gating signals.
The aignal is divided by 2 by E22 $(8,9,11,12)$ and again by 2 by $:$ E22 $(2,3,5,6)$, and then by 16 by E23. By gating these signals together, a set of waveforms shown below is produced.


SIGNALS A1, AO, BO, CO provide a four bit address code to the three data multiplexers. The top two multiplexers, E15, E27 are connected to the keyboard, and as only 30 notes are provided, then two inputs (E27 pins 7,8) are left permanently unconneoted. The lower multiplexer is the "transposition pad" unit.

MULTIPLEXERS - an electronic 16 way switch; single pole.

$\therefore 8$
The 16 inputs; EO to E15, can be considered as the 16 inputs to a rotary switch. The switch position (only one pole) is determined by the address code, ie AO, A1, B1, C1, labelled $A B C D$ on the actual device.
$\overline{\text { Switch }} \overline{(A} \cdot B \bar{C} \quad \bar{D})$
Position AO A1 B1 C1


The ENABLE SIGNAL acts as a switch; that is when this signal is high (1) the switch is open circuit, and the output (w) is permanently high.
When this signal is low ( 0 ) the switch is closed and so anyone of the 16 data inputs may be sellected. The ENABLE'IS SOMETIMES called the STROBE.

Note that the output shows the inverted data. Thus to summarise. With a high (1) at data input 12, and address code of $\begin{array}{llll}A & B & C & D \\ 1 & 1 & 0 & 1\end{array}$ and an enable signal of logic 0 , the output is (1) inverted, namely 0 .

## TOUCH SENSITIVITY.

The input to a TXI gate looks like:-


This is a multiple emitter, transistor, i.ee. a transistor with, in this case; 6 emitters.

In the multiplexer, 4 of them are connected to the address code, one to the enable signal and the last to the touch plate.


For ea ch specific input there will only be one point in the Address cycle ( 32 bits) when all 5 emitters are high at the same time. When the bias is correctly aligned, the circuit will behave as follows:-

looking at the last trace we see that sixth emitter, which is connected to the touch plate, is dragged up, due to the vil.
associated capacitance of itself WRT its sister emitters. Infact, it is dragged
up so. far that it changes state from a ${ }^{\prime} 0^{\prime}$ to a '1'. BUT when a finger is placed on the associated touch plate, the: capacitance of the finger increases the total capacitance as seen by the emitter, increasing it by as much as 10 to 100 pF , and this is sufficient,
to prevent it from changing state. Thus, pressing the pad with a finger
 will prevent the data multiplexer output from changing state.

The enable signals ' $M$ ' and ' $N$ ' are


Thus one multiplexer is addressed, and then the other (i.e. E15 and 27) is then the other etc.

## op $W$ of $E 27$



If more than one note is pressed, then several gaps in the outputs will be generated; however; the electronics that follow select the first note to appear, and as the keyboard is swept (addressed) from top to bottom, then the highest note of all the notes pressed is selected.

This piece of electronics is $\operatorname{E26}(11,12,13)(4,5,6)$
E25 (8,9,10,11,12,13), E16, E17, E18. I :
The two data outputs of MULTIPLEXERS E15, E17 are OR' ED together by E26 (11, 12,13) (that is they are effectively just added together).


*WHEN looking for these waveforms, you much SYNC OFF of DO E23/11.

Now consider the ' $D$ ' type $E 25(8,9,10,11,12,13)$.
Assume that pin 8 is at logic 1. Therefore, Pin 5 E26 is at logic 1, and it is thus "enabled" in that is allows signals at its other pin (pin 4) to pass through it. The signal at pin 4 is the summed data output, which when it passes through the gate, is inverted. Notice that the clock pulses are coherent with the data signal.


The D type operates as follows. It is an edge triggered device; that is data at its' $D$ input is transferred to the $Q$ output on the +re transition of the clock. That is, put a 1 on pin 12, clock the device and a 1 appears at pin 9 , the $Q$ output. SO, in the present example, when there are no notes pressed, the data is low when it is clocked, and so $Q$ remains Low. But, when that data is high, then $Q$ goes high, therefore $Q$ (pin 8) goes low and in doing so inhibits E26 $(4,5,6)$ the output of which thus remains high. The circuit has latched up, and it will remain latched up until it is
 released by a clearing pulse. Thuse, E25, Q9 goes high and remains there when it detects the $\mathbb{I}$ first note. (See page 2, waveforms). Now, E25, Q9 clocks the $D$ types $E 16,17,18$, which are fed with the address code ( $A, A_{0}, B_{0}, D_{0}$ ). So, as soon as the first note is detected, its address is clocked into E16,17,18 which act as a one bit memory. At the end of one address sweep (about log sec.) the $D$ type $E 25$ is unlatched by a negative going pulse produced by differentiating the address bit $D_{0}(23$ pin 11) with the CR network C4, R44.


A Low on the clear input of a D tape cleans the $Q$ output to logic $O$

Thus, the $D$ types E16, 17, 18 form a short term memory that is continually being updated (every $100 \mu \mathrm{sec}$ ) as to highest key that is being :ier: selected. The output Pin 9 E25 is an excellent place to test the keyboard to see if the multipexer for the KB and/or the address code is functioning.


As you sweep your face up the keyboard
The waveform chances as show.
(The pulse width elarcates)

RANDOM FUNCTION This is the same touch sensitive detector as for the KEYBOARD. However, as there is only one input, then no addressing is required. A slow clock pulse generator is used, ( $5 \mathrm{KHz}, 60 \mathrm{n}$ sec. pulses). A separate'oscillator, rather than the address oscillator is used, because:-

1. A long enough gap must be provided to allow the floating gate time to discharge.
2. If the address clock were to be used, the Random would not be random; this is because the network has a tendency to select the largest clock pulse and this means that particular notes occur more readily than others.

The operation is as follows. When a missing pulse is detected by E25 pin 2, the $\bar{Q}$ output goes low and so the cct folds over. The $\bar{Q}$ is connected to the set input of E2S. This will cause the $Q$ output to go high and in doing so it ' records ' an address. (Any.address in fact; it depends upon when the Random is pressed.) . When the random is released, the set input of E25 pin 10 returns to 1, and the circuit operates as before. As the $Q$ output $E 25$ pin 6 can go low at anypoint in time, then any note is selected, and hence the term ' random '. N.B. The random function can generate errors; this is a design error.

The errors can be generated by the random accidentally selecting address code Binary 11111 ( i.e. 32 ). This is the ' Skip ' code, and it will cause the meter to prematurely reset, and the memory to corrupt. To demonstrate this phenonemum, try this test; Set the Sequencer clock speed to about 3 to 4 on the knob, press record and then tap the random pad. There is a $32: 1$ chance of generating the skip code, and this will be detected by a premature reset of the meter. DAC

The short term memory $E 16,17,18$ feeds Q1 to 5 which form a DAC. ( Digital to Analogue Converter ) Slightly re-drawing the circuit, we the following;

Inside the dotted box are 5 resifstors and 5 switches. The resistors are R49 to R54, 1\% m.o. resistors. The switches are Q1 to 5 and these are controlled by th stored address in E16, 17, 18.
When the address is 00000, all the switches are off, and so the circuit is just a simple differential amplifier, and the voltage at the output of A1 is OV. When the switches turn ' ON ', the output goes negative.


- NOTE URT the short term memory E16,17,18. The lower half of E18 is different to the other 5 sections. This is..the "key'pressed" section. It uses the clock line as its data input. This is because, when Q9, E25 is high, then a key is being pressed, and this is the data that it requires. It is clocked by $\bar{D} 0$ (known as $\overline{\mathrm{F}} 8$, E26) at the end of an address sweep, and so its $Q$ output (i.e. 9, E18) is high when a key is pressed and low when no key is pressed.

Note that the negative (1-0) transition of the $\bar{Q}$ output ( $8, \mathrm{E} 18$ ) is used to control the envelope trigger by turning transistor $Q 6$ on when a key ispressed. The envelope trigger control pot is used to determine whether the trigger is generated from the real time or recorded output.

DYNANIIC RESFONSE The crystal gives the K.B. a dynamic function; but this is only employable in real time. That is, no information about the key pressure is recorded. The crystal produces a click when the K.B. is hit.


The network is a voltage follower with a diode in the feedback loop. The junction between the diode and the capacitor tries to follow the tye voltage transient of the input and decays slowly (due to the storage of the capacitor) after the signal has ceased. This voltage is used to control one of the output channels on a Synthi A.

RECORD, PLAY, CONTROL IOGIC * note RECORD AND PLAY are clocked from the random clock gen.

The RECORD and PLAY pads operate in the same way as the Random touch pad. However, the interconnection of E10 (1,2,3,4,5) with E10 ( $8,9,11,12,13$ ) and E10 with E18, provide a sequential mode of operation. That is:-


$$
\text { RECORD } \rightarrow \text { KEY BOARD } \rightarrow \text { PLAY! }
$$

$$
\begin{aligned}
& \begin{array}{l}
\text { IO }\left\{\begin{array}{ccc}
Q 6 & \frac{R E C O R S}{} & \frac{\text { TOUCH }}{\text { KB. }} \\
Q 8 & 1 & 1 \\
\text { Q } & 1
\end{array}\right. \\
\text { ASSUME EI8/8 is HGH }
\end{array} \\
& \text { E10/6 is HGiA } \\
& \text { E } 10 / 8 \text { is HGA }
\end{aligned}
$$

Remove finger - Allows Sequence to Play. Pressing' Puma' nareur stops Sequence.

Reverts back to a 1 when finger is recoaseo
3. Play. is pressed. A logic high is presented to the $D$ input (12, E10) and so the $\bar{Q}$ goes low. However as there is no...

Sa. latch up it returns to its high level as soon as the finger is released. Note, that when at the record mode, the clear input (13, E10) is low and 50 the lower half of $E 10$ is prevented from changing state; that is, Record Play does nothing.

Control Logic This is E6. The logic sequence (Above) is presented to E6. The output sequence generated by E6 is as follows:-


Lines $A$ and $B$ open and close the sets of input/output gates. Only one set of gates (A or B) is ever open.
When the logic level on $A$ is high, gates connected to it are open.
When the logic level on $A$ is low, gates connected to it are closed.

i. KEyB'D Mode. This is "record" mode (confused?); data is accepted, because $A$ is high. The data is clocked into the shift register at a slower rate, this rate being variable. If you are unlucky, you can overfill the shift register, and the front of the data is merely lost out of the end of the device.

Play. Data is replayed and recirculated; because line B is high. The replay clock rate is also variable. When an unfiffed part of the shift register is detected a skip mode is implemented and the shift register skips all the blank locations until it finds the start of the data, which it then plays at its slow rate. Pressing the play pad stops the shift clock from operating in its "play" mode and so pressing play whilst the machine is replaying does two things:-

1. Effectively pauses the device
2. Upoh releasing finger, the sequence restarts (proc'eded by a skip block). If the shift register has been overfilled, then there are no blank locations and 50 when the play is pressed, the machine rapidly searches for an empty location and cannot find one. The result is a buzz sound as all the data rapidly recirculates. Also, as there are no blank locations, the machine does not know where the sequence start is. Therefore overfilled sequences, when replayed will start anywhere.

The memory is a 6 bit word, 256 bits long. 5 bits are used to record the note selected. The sixth is used to trigger the envelope trigger; that is if. the envelope trigger pot is in the right position. SHIFT CLOCK/SKIP GENERATOR

E19, Q14, C6, R71, R70, R72, VR3 is the high speed skip generator. It is a schmitt trigger oscillator. When C7 is operational the oscillator reverts to a low speed clock generator. In normal operation pin 8, E3 is low, C7 is thus effective and so a slow clock is generated. (Normal operation is during the recording and playing back, and not the pressing of those pads).
" In normal operation, the output of $E 2$ is high (i.e. the input is not all 1's) and so is the input pin 9, E3. When RECORD is pressed, all the inputs to E 2 are high, the output of E 2 goes low, and sets the skip oscillator into its skip mode, by making C7 ineffective. The shiftregister is thus filled with O's. Also, the two meterdivider circuits E2O, E24 are reset to zero.

When the KEYBOARD is pressed, the input gates to the shift registers are opened, and so the inputs to $E 2$ change from all 1 's. to a mixture of 0 's and 1 's' (Note an all 1 's condition cannot occur in normal operation). The output of E2 goes to 1. . Thus, the output pin 8 of E3 goes to 0, C7 is functional so the oscillator begins to clock in its slow mode. Also, the reset on $\mathrm{E} 24 ; 20$ is in its count mode, the clock being taken from E19 pin 8 . The output from E19, 8 is differentiated by C16, R68 to give a very short pulse, which appears at Q13's collector, and is used to drive all the shift registers.


When the PLAY pad is pressed. . Input pin 9, E3 goes low (but only for the duration of the pressing of the pad); this causes the data to rapidly recirculate (skip), usually for one'bit only (because E3, 8 is high).

$$
\begin{aligned}
& \text { ES, } 9 \text { is low. Hence. } E 19,6 \text { is high. } \\
& \text { ER, } 8 \text { is high. }
\end{aligned}
$$

Nothing happens until an "all low-state"e of the shift registers is detected by E2.(*all 1's as seen by E2). When this state is located, EL, 8 falls to 0 , therefore. E3, 6 goes high, and in doing so resets E2O, E24. Also, there are now two high's at the input to E3, $(1,2,3)$ and $s 0$ ES, 3 goes low and in doing so disables E19 ( $8,9,10,12,13$ ) the oscillator. If there are no empty storage locations, the network just rapidly recirculates, and a "buzz" is heard.

When the play pad is released, $\operatorname{pin} 9, E 3$ goes high, however E3, 8 still remains high; E19, 6 takes a short time to go low because of the effect of C17: (This eliminates spurious effects when operating the play pad.) When E19,6 goes low, E3, 3 goes high, thus enabling the clock oscillator. The clock is in its high speed mode (skip) and so it races past all the unused storage area and when it comes to the first piece of data, E2, 8 goes high, the clock thus reverts to its slow speed mode, and E24, E2O is set to its count mode. Thus, the stored data is replayed, and when the end is reached there is a rapid skip and the process repeats itself. METER COUNTER

The clock is dived by +16 twice, which is 256 , the length of the memory. The 5 most significant bits are decoded in the meter DAC. This is E24 and the resistor network. The network is a "weighted/ladder" - (a Cockeral special).

MSS



The biggest error in Linearity bemas $\frac{1}{2}$ way
WHIERE A PRonounces ster CN SOMETiMEs occura.

 of the address bits.

The switches are transistors $Q 7$ and $\& \delta-Q 1$ ? and the operation is similar to the real time DAC. Note, that the transposition acts only on the recorded output DAC, but no record of transposition playing is stored in the memory. Trans:osition is real time only.

## TRANSPOSITION.

The transposition miltiplexer operates in a similar manner to the two other multiplexers. One difference is a feed-back circuit $\mathrm{E} 11(8,12,11)$ that causes single events to be multiplied. Looking at E7, we have:-


When Eo is pressed, it is prevented from going high, and so the output at $W$ is high for Eo.
therefore pressing Eo produces a 1 bit pulse at E11 pin 8.


Pressing the tone causes $W$ to be high for E1. But the output $\bar{Q}$ of E11 is fed back to E2. And so when E2 is addressed the output is again high but stops when it gets to E3.
; :

-Notice E14, E15 are never low.

Now, the output from E.11 pin 8 is used to oontrol the DAC on the recorded output. . As the transposition signal is a mark/space ratio signal, then the resultant product from the DAC must be smoothed out, this being accomplished by C5. The transposition signal is used to control a logic switch of equal status to the MSB of the 5 bit DAC.

The thirty notes available from the keyboard form $2 \frac{1}{2}$ octaves, (i.e. 12 notes per octave). Therefore 12 logic bit $=1$ Octave

```
or . : 1 semitone = 1 logic bit.
```

where 1 octave = 12 semitones.

The MSB of the address is equivalant to 16 logic bits. Therefore as the transpose controls a switch of equal status to the MSB, then one bit of the transpose signal $=1$ semitone.

Therefore, pressing the $\frac{1}{2}$ tone pad $=1$ bit $=1$ semitone.
pressing the tone pad $=2$ bits $=2$ semitones.
pressing the third pad $=4$ bits $=4$ semitones.
pressing the fifth pad $=7$ bits $=7$ semitones.
that is, pressing the keyboard and the transposition.pad, tranposes the note generated upwards by the respective amount.

* Example:- pressing the $\frac{1}{2}$ tone, third and fifth will shift a note up by one octave. (That is, if the keyboard has been adjusted to $2 \frac{1}{2}$ octaves)
$1+4+7^{\prime}=12$ semitones = 1 octave.
(* An octave - the ratio of two frequencies which are at frequencies of $F$ and $2 F . \quad F$ being arbitary).

APPENDIX


1 .
D type, used usually as a one bit delay or an edge detector.

Logic data is presented to the $D$ input. The data ( 0 or a 1) appears at the Q output after the +ve edge of the clock pulse, that is:-
data is transferred to the $Q$ output on the +ve edge of the clock:-
Set - a 0 on the set puts $\mathcal{E}$ to 1
Clear- a $O$ on the clear puts $Q$ to 0
set and clear are independant of the clock and overide the $D$ input. SCHMITT


SNTH13N

The SN7413N looks like a 4 input AND, followed by a schmitt trigger, followed by an. inverter. The schmitt trigger exhibits a phenomena known as Hysterysis.


## AKS Patch Sheet




Hi-Zo drive amplifier. . . Pick-up amplifier a voltage controlled mixer.
$12 v$








$\qquad$



Anti-log fraquency Integrator. Hysteretic switch D.C.shifter. Rectifiier. Square wave output.






## SERVICEMANUAL <br> Synthi A/VCS3 <br> DK1/2 keyboard <br> KS sequencer

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Telephone 0726-883265
We reserve the right to make changes in technical specification and design in the interest of continuing improvement in performance.

CONTENTS:
Section 1 Synthi A/VCS3
Section 2KS keyboard
Section 3 DK1 and DK2 keyboard

NOTE:

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Transistor types used in cur machines have varied from time to time.
The following types are the most widely used:
NPN sil BC 169C 2N 5172
PNP sil BC 258B 2N 4288
Fast NPN sil ME9002 2N 706
Fet 2N 5163
Fet 2N 5461
Diodes:
DK10 - small signal germanium
1N4148 - 1N 914
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## SYNTHI A ALIGNMENT

Synthi A
Testing Manual Board

1. Power Supply ..... A
2. Output Channels ..... A
3. Reverberation ..... A
4. Envelope Shaper ..... B
5. Filter Oscillator ..... B
6. Ring Modulator ..... B
7. Input Channel Preamplifiers ..... B
8. Oscillator 1 ..... C
9. Oscillator 2 ..... C
10. Oscillator 3 ..... C
11. Noise Generator ..... C
12. Meter Amplifier ..... C
;
Board $A$ is on the left side as you look at the opened back of the machine. Board $B$ is in the middle, with Board $C$ on the right.

## SPECIFICATIONS

Power: $100-130 \mathrm{~V}$ or $200-250 \mathrm{~V}$ ac.
Stereophones level: 10 V p.p max into 50 ohms
Scope output: 5 V p.p $\max$
Signal output: 2 V p.p max into 600 ohms
Control output; 15 V dc max into 10 K ohm
Signal Input; ac -1.8 V p.p into 50 K ohm. dc- 2.5 V max into 50 K ohm.
Microphone Input; 5 mV into 600 ohms

## Devices



| Envelope Shaper | Duration: Attack $2 \mathrm{~ms}-1 \mathrm{~s}$  <br> On $0-2.5 \mathrm{~s}$  <br>  Decay $3 \mathrm{~ms}-15 \mathrm{~s}$ <br> Off $10 \mathrm{~ms}-5 \mathrm{~s}$ (+Off position) |
| :---: | :---: |
|  | Trapezoid Output: $\pm 3 \mathrm{~V}$ dc max. Decay Control; 0.4 Volts/octave |
| Reverberation | Max time: 2 s |
|  | Output Level: 5V p.p |
|  | Mix VC : -2V for 0\%, +2V for 100\% mix |
| Joystick: | Voltage Output: $2 \mathrm{x} \pm 1.5 \mathrm{~V}$ dc |
| Meter Input | Control Voltage mode: $\pm 1 \mathrm{~V}$ dc |
|  | Signal mode: 4V p.p |
| Input Amplifiers | Sensitivity: 1.8V for High Level Inputs |
|  | 5 mV for Microphone Inputs |
| Output Amplifiers | Output level: 2 V p.p into 600 ohm (panned) |
|  | $10 \mathrm{~V} \mathrm{f} . \mathrm{p}$ into 50..ohm (unpanned) |
|  | 5 V dc into 10 K ohm (unpanned) |
|  | VC level: $15 \mathrm{~dB} /$ Volt |

1 :



## Power Supply

Before turning on, set PR1 cw and PR2 ccw with only A board plugged-in.
Set PR2 for $+12 \mathrm{~V} \xlongequal{2} .1 \mathrm{~V}$
Set PR1 for $-9 \mathrm{~V}=0.1 \mathrm{~V}$
Check PR2 again and reset if necessary
N.B. -9 V rail is parasitic on the +12 V supply.

Plug in $B$ and $C$ cards and recheck $+12 V$ and $-9 V$ rails.

These should not have altered. If they have, suspect either B or C board or poor regulation on A board. Plug-ni keyboard as extra load. +12 and -9 Volt rails should not have moved more than $\perp 0.1 \mathrm{~V}$ measured always on the A board. ( C board voltage may be 20 mV lower)

Monitor the rails on the scope ( $5 \mathrm{mV} / \mathrm{cm}$ ). Look for hum and noise. These should be much less than 5 mV . Noise, if present, will probably be due to the zener diode.

Power supplies should stabilise on full load for specified fluctuations. $110 / 240$ nominal : $+10 \%$ or $-23 \%$
110 nominal : 100V-135V for 125 sufficient
240 V nominal: 200V-260V

Hum can also be checked by the following:
Check Output Filters on 10
Output Controls on 10.0.10.10
Patchboard vacant
Monitor ( $5 \mathrm{mV} / \mathrm{cm}$ ) on signal output
Load unit with keyboard
100 Hz ripple as shown.
Noise can be seen which should never exceed 10 mV p.p. Even with noise level up. With noise level down, the noise on this waveform should be quite low. Zener diode noise will be very obvious.
N.B. A2 normally gets hot, dodgy D1 gives large 100 Hz ripple on Output. Always clean edge-connectors and check clean presets on $A$ board.


ALL $2 N 5172$ NOW BC169C


## Output Channels

Set Output Channel Filters to 5.
Set Output Level row to 0.0 .10 .0 and connect traces of scope ( $5 \mathrm{mV} / \mathrm{cm}$ ) to both signal outputs.

Connect full level ramp signal at 261 Hz from Oscillator 1 to both Outputs. Trim PR4 and PR5 for minimum signal breakthrough. (Maximum 5mv p.p) Replace presets if adjustment is erratic.

Reset Output Channel row controls to 5.0 .10 .5 . Output levels should be around 1V p.p and should be within $20 \%$ of each other. Check signal tracking in the same manner for Output settings 4 and 6 as well. If one level is consistently high, pad out resistor across its control pot (R305 $=56 \mathrm{~K}$ )

Successful tracking will depend on the characteristics of the pots and the correct matching of the FETs Q14 \& Q18.

Connect the joystick row via two shorting pins to both VC level inputs. Swing stick up and down to check for $20 \%$ levelf tracking tolerance. (Level pots should be at zero). Inaccuracies in stick tracking may be adjusted by altering the value of R302 or R304 (10K) which are mounted under the patchboard rear.

Check correct operation of $\operatorname{Pan}$ controls, which only affect the Signal Outpists. Check also the Output Filters. Use noise as a gcod source.

Check for max. outputs on the Signal Output socket of at least 2.5 V p.p before the onset of clipping. Level controls should be at about 6. Check
 hard with low frequency. (Oscillator 3 square wave). The level control of the channel NOT beirg driven should be on zero. Monitor the two signal outputs. The undriven channel will show breakthrough. Plug in headphones and check for oscillation. Reduce drive to Output Channel 2. Keep checking for oscillation. Then repeat for other Output Channel.
N.B. DC offset across lcudspeaker is $1.8 \mathrm{~V} . \pm 10 \%$.
.Hi-Zo drive amplifier.
Pick-up amplifier a voltage controlled mixer.


## Reverberation

Set PR3 fully clockwise before operating. Feed large amplitude tone bursts into reverb. Adjust $\operatorname{PR} 3$ so that the output is dry at Mix setting of zero and reverberation progressively increases from setting of 2.

Maximum dryness on the Mix control should be no worse than that obtained with the VC Mix input row on the patchboard grounded.

Check voltage control of Mix with the joystick.

Hold input amplitude constant and sweep frequency of tone. Check for rattles, buzzes or abnormal resonances in the spring. Check spring is correct way round. If Mix control is wrong change Q10.
N.B. Reverb amplifier should clip symmetrically.

Q45 3.5-4.0 A1. 2NS172 NOW BCI6gC
048
$O R>1.5$ ADJUST R143


## Envelope Shaper

Put scope on $2 \mathrm{~V} / \mathrm{cm}$ and connect to Trapezoid output. With envelope controls at 0.0.0.0. Adjust PR12 so that frequency of trapezoid oscillation is about $90 \%$ of the maximum possible frequency. Connect Filter output on 261 Hz (Frequency setting 5) with maximum output to Envelope Input. Monitor Envelope output. Set PR13 clockwise.

Set envelope controls 0.0.0.10. Adjust PR7 for minimum signal. Adjust PR13 until signal reappears, then back off slightly. The finalenvelope shaper signal breakthrough should be 5 mV or less p.p. With the Attack button pressed the output signal should be greater or the same as the input signal.

Set controls 0.10.0.10. this gives maximum On time. Press Attack button which should recycle the envelope shaper and lamp should come $0 n$, staying on for about 5 seconds. (absolute minimum 4 seconds). This corresponds to an automatic recycle On time of 2.5 seconds which is the minimum spec. (On time is always shorter if recycled automatically). On time is controlled by R148 in conjunction with Q45. $\}$ :

Set controls 0.0.10.5. Press Attack button and time the Decay time. This should exceed 15 seconds, and the envelope shaper must automatically recycle at the end of the decay.

Set controls 0.0.0.7 to 10 . Envelope shaper must not recycle with the Off time on 10 , so check it doesn't on 7 or so, which will indicate the behaviour on 10. Check input breakthrough at the same time, and adjust PR7 again if necessary.

On and Off times are controlled by R148 (range 10-18K). If On time is infinite at 10 , reduce R 148 , if off time infinite, increase R148. The value of Vp for Q 45 (3.5-4.0 nominal) is critical. Failure here upsets recycling. Also leaky C48 or dodgy envelope trigger components .


## Filter/Oscillator

With Res:ponse on 10, level full and Frequency on 5, check for oscillation. About 1.5 V p.p should be obtained. The shape should be reasonably sinusoidal especially when only just oscillating (on Response control).

Distortion should be eliminated either by diode bridging or by replacement of offending diode. Bad diodes can be detected by applying the tip of a soldering iron quickly to the diode body. Allow enough time for the diode to cool down before moving on to the next one.

Response control should make the filter oscillate only above 5 for all frequency settings. If this does not occur correctly, adjust R90 or suspect wrong Vp FET. Frequency range on the Frequency knob should be roughly 10 Hz to 12 kHz . If oscillations stop at one extremity of Frequency pot increase R75. Adjust PR6 for oscillation frequency of 261 Hz with $F$ on 5 . Noise level of Filter output signal with Response set below point of Osc should be below 5mV p.p.

Check correct filtering using square wave input with swept $F$.

Intermittent noise: Try soldering iron tip on diodes. R288 = 1 megohm stops Filter latching-up.


## Ring Modulator

Use filter sinewave output, $F$ on 5. Maximum response, maximum level into both inputs of ring modulator. Ring Modulator level maximum. Check output is double frequency of input, and at least as big. Remove pin to Input B. Drive input A and adjust presets for minimum breakthrough. (PR9 - second harmonic rejection, PR10 - fundamental rejection).

Drive Input $B$ by removing pin to Input $A$ and reinserting pin for input B. Trim presets for rejection. (PR8 - second harmonic, PR11 for fundamental). This one should NOT drift. Input A drifts quite a bit and should ideally be adjusted with the back on the machine after it has fully warmed-up. If this is not possible ensure that Input $B$ is used for the continuous signal to minimise breakthrough.

With an input level of 1.5 V p.p up to 10 mV p.p maximum is allowed for breakthrough. This gives about 60 dB rejection.

If higher harmonics are a problem, then filter signal must be too impure. Try backing off response, or use an external signal generator. 1.5 V p.p at 261 Hz .

If reasonable rejection is unobtainable, change matched pair BC169C or TAB101 where appropriate. 2 C 746 used on later models. Do not bridge with resistors. Replace presets if adjustment is erratic.

CH.II.


Input Channel Amplifiers


Using a signal generator check the two inputs for the expected gains.
This is best done by monitoring the control output with shorting pin. High jevel gain can be cinecked by appiying about 2 V p.p and looking for $4 V$ at the output.
Low level (microphone) gain : 50 mV should give 5 V p.p.

Check for minimal DC offset by plugging input channel inot the meter switched to Control Voltages, and then rotating the input level pot. No more than $\pm 1 / 2$ a division movement should occur. (approx 200mV).

Pay special attention to condition of pot tracks.


## Oscillator 1

The first operation is to adjust the shape of the sinewave, since the adjustment affects frequency. Using the vernier control and PR15 if necessary, obtain an output of 400 Hz . Turn the ramp down and adjust the sine shape control for the best sine shape. Trim PR17 to balance out the spike on the peak of the sinewave. This may best be seen by expanding the $X$ and $Y$ inputs on the scope.


The best position is shown, where the small spike is triangular and balanced. A small thin spike may also appear but is not important.

Test the main shape control. A good sinewave should be acheived on 5 . ( $\pm 1 / 2$ division). If further out, R197 or R201 may, nęed to be adjusted. Set PR16 halfway and leave it alone. Set frequency vernier dial to 6 and adjust PR15 for 261 Hz . Rotate vernier to 8 and check for 2088 Hz . Absolute mazimum error $\neq 1$ division. For fudge procedure, see Osc 2 writeup.

Sine Output should be around 1.5 V p.p
Ramp Output should be around 1.5 V p.p
N.B. Ramp shape is not controllable.

Pitch characteristic is 0.32 Volts/Octave.

Waveform Shint A ADust


## Oscillator 2

Set square level to zero, ramp at maximum. Set PR20 halfway. Vernier on 6 . Trim PR19 for 261 Hz . Increase vernier to 8 . Check frequency is about 2088 Hz . If beyond spec then R291 SOT will have to be added in the range $8-22 \mathrm{~K}$. Check that 10 on vernier dial gives at least 10 kHz . If vernier dial at 8 gives less than 2088 Hz then short circuit R291 and R213 will have to be reduced. These should be $2 \%$ resistors. Shape controls should operate identically to those on Osc 3 (see next page) except that levels are reduced, and the two outputs are summed into the patchboard on Mark 11 machines. Remember to turn down unwanted signal. Resistors R226 and R228 are the relevant ones to set the shape. Pulse level should be 2 V p.p. Ramp level should be 3.5 V . N.B. Osc2 and 3 waveforms are not symmetrical with respect to zero volts. Tracking

Mix Osc 1 and Osc 2 ramp outputs into Output Channel 1. Use PR20 only to trim the tracking of Osc 1 and Osc 2. Using the joystick on max with range control to limit frequency range. Set both verniers to 6 and tune for zero beats. Advance the joystick range conthol which will increase both frequencies until the beats begin. Trim PR20 to zero beats, and remember which way you turned it. (N.B. The preset is a very fine adjustment, and when the tracking is nearly right it will be necessary to make very very small movements). Having trimmed PR20, turn the range control back to zero, readjust one of the verniers near 6 to give zero beats and then advance range control of joystick to where it was before. If beats are faster you know you turned the preset the wrong way. Repeat procedure until tracking goes to about 2 kHz . Ideally a test 'back' should now be fitted and a final adjustment made after another 10 minutes have elapsed.

## Faults

Tracking from 261 Hz to about 2 kHz should be achieved with a maximum of 3 beats per second on the way. A curved Voltage/Frequency response may occur where zero beats are produced in more than one place. This is due to a mismatch in the matched pair transistors (Q74 \& Q75 for Osc 2). Fit a new transistor here or in 0sc 1.
N.B. Always use specially matched 2 K 7 pins for tracking tests.

WAVEFORM-HAPE ADJUST


## Oscillator 3

Check frequency control pot for even change in frequency.
Adjust PR21 so that when the vernier is at 8 the frequency is 63 Hz .
Note: Vernier 10 should give 500 Hz . Period at 0 should be greater than 20 sec . Adjust values of R261 \& R262 with pad resistors to achieve correct shapes of waveforms as shown below:

## 6 V php $1 / \mathrm{N}$ <br> Shape at 0

5V pep



Shape at 10

Triangle and square wave should occur on 5. Maximum of $1 / 2$ division error allowed.

Check Voltage control input.
Faults
Osc 3 breaks through to Osc 2 and to a lesser extent Osc 1. Effect dependent on Osc 3 shape control setting. Triangle was 'furry', with 10 mHz oscillations visible ali over C board. Strongest near Q101. Cause: instability in Q99. Q100, Q101. Cure, C70 330pF change to 1 nF .


Ramp defects


Ramp defects


Pulse defects at 0 and 1


## Noise Generator \& Meter Amplifier

Colour control on 5, max output. Select a noise diode which gives wideband noise and can be adjusted to give $3 V$ p.p with PR22 roughly midway (to give some later adjustment margin).

Lumpy or crackly sounding noise diodes should be rejected.

## Meter

Check left hand zero of meter with machine off or switched to Signals. Switch to Control Voltages and adjust PR14 for accurate centre zero. Switch the meter switch a few times and recheck zero again. Patch the joystick to Meter for swings in both + and - directions. Check for needle sticking at either end of travel. Switch to Signals and test with Osc 2 sawtooth at max level. Vary level to check proportionality.

## Faults

Watch out for the resistor R170 on the C board, which often touches the -ve end of C76, and has been known to produce curious failures on soak test.

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MEIEK AMRLITIEK.
D.C voltage follo

Rectifier.
$+12 v$ IOISE GENERATOR \& METER AMPLIFIER $\quad$ P.C.BOARD " " "




INPUT CHANNEL AMPLIFIERS P.C.BOARD "B"

vcs $3 / 15 S 2 /$ CCT 7







## Dynamic Keyboard - Setting-up procedure

Set all knobs and presets halfway.
First set the keyboard zero volts at middle F sharp. Press middle F sharp with scope on Pin 13 of keyboard circuit. Adjust PR3 for 0 Volts $\pm 50 \mathrm{mV}$. Check that top $C$ gives $1.5 \mathrm{~V} \pm 200 \mathrm{mV}$
Check that bottom $C$ gives $-1.5 \mathrm{~V} \simeq 200 \mathrm{mV}$
Check drift rate when key is released. Should be less than 100 mV per minute. If greater suspect Q12. This should have a Vp of 2.5 - 3.0 .

Keyboard Oscillator
Press middle F sharp. With VR1 (Frequency) at 5 and VR3 clockwise put scope on live connection of jack socket 'Keyboard Modulator'. Adjust PR1 for 248 Hz at F sharp or 261 Hz for middle G .

If PR1 is off range clockwise then snip R9.
If PR1 is off range counterclockwise then reduce R9 value by half.
Hold down bottom $C$, and set !'R2 (Tuning) to 5. Press C one octave higher. Trim PR2 to get exactly an cctave higher than bottom $C$.
If PR2 is off range clockwise then snip R11
If PR 2 is off range counterclockwise suspect cirduit fault.
With Frequency at 0 press lowest note. Should be $\because$ lower than 30 Hz .
With Frequency at 10 press highest note. Should be higher than 8 kHz .

## Dynamic Voltage

Short across C9 and trim PR4 for +3 V at Pin 12 on keyboard circuit. DO NOT PRESS A KEY WITH SHORT IN PLACE.

Check for drift, which should be less than 100 nV in ten seconds.
Check that Dynamic Range has no effect on keyboard oscillator loudness when turned to zero. If so re-adjust PR4.

## Contact Displacement

Each note has two sets of contacts. One pair, that is not connected to the resistors, must be adjusted to make about . $03^{\prime \prime}$ after the other pair. With practice it is possible to do this by eye using a fine pair of tweezers. With VR1 and VR2 halfway, VR3 and VR4 fully clockwise and both switches up. Monitor the signal output on a scope. Strike each note with a even force. This should give about 10 dB less ( $1 / 3$ as much) than hitting it hard. Maximum level should give about 2 V p.p on the scope. Adjust the crooked member of the late pair where necessary. Closing the gap makes the note louder. If

Dynamic Keyboard - Contact displacement (continued)
visual alignment still gives performance inconsistencies this is almost certainly due to contacts being dirty. This is a frequent source of problems on the keyboard generally, giving swoops and dithering pitches at different places on the keyboard. Use cotton buds and Isopropyl alcohol to carefully wipe each surface of each contact. Extreme care must be taken not to disturb the contact displacement, but this will restore precision to the keys. Never use an abrasive substance on contacts since they are gold-plated. Blistered plating means new contacts must be fitted.

## Setting up DK2 keyboard

DK2 keyboards are duophonic, so that Channel 1 pitch voltage comes from the lower note being played, with Channel 2 pitch voltage coming from the upper note. This second upper note voltage is derived from the lower note and PR6 needs to be adjusted to get the relationship right. Connect Channel 2 Voltage to control an oscillator with the Input Channel gain at maximum. Hold down the bottom key with sellotape and play the top note. If the oscillator pitch shifts adjust PR6 until there is no more movement. Some keyboards were modified from DK1 (monophonic') to DK2 bY the addition of a small pc board under the main one. In these cases PR6 is the preset on the small board.


FIRSTLY. The Touch Keyboard and address clock.
The address clock is an oscillator, E28 (Pins 1,2,4,5,6), being a conventional schmitt trigger oscillator.

OPERATION Pin 6 is high; D38, 43 are off, C1 and C1X charged up by way of Pins $1,2,4,5$ until upper threshold voltage (about +1.7 v ) is reached, the schmitt then fires, Pin 6 goes low, D38, 43 turn on and discharges C1 and C1X until the lower threshold voltage (about 0.9 v ) is reached. The schmitt then reverts back to its original state etc.etc.







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c. Rus ros. isvis 171 $+-3 \cdot 73$


Cin 116 inve. 23.7.74




