



## 2. Original Circuit description of Audio Processor Block form T881 service manual

T881 Circuit Operation. In the service manual the corresponding part is 2.5. The circuitry of sections 2.5.1, 2.5.2, 2.5.4 and the first paragraph of 2.5.5 have been functionally replaced by the DSP via the modifications described earlier. The sections (marked \*) 2.5.3 and last two paragraphs of 2.5.5 still apply.

### **2.5      AUDIO PROCESSOR**

#### **2.5.1      GENERAL**

The audio processor comprises several link selectable circuit blocks, which may be configured in a variety of combinations to suit individual requirements. The pre-emphasis network and compressor may be linked individually or cascaded between either or both audio inputs and the limiter.

Refer to Section 3.5.1 for linking details.

#### **2.5.2      AUDIO INPUTS**

Two audio inputs are available: one from a 600 ohm balanced (or unbalanced) line, and the other from a local microphone. The microphone signal is passed first to a pre-amplifier (Q210) and ultimately to a multiplexer (IC 1240), but in between may pass through the compressor (depending on the linking details). The line transformer is also connected to the multiplexer and is disabled by the microphone PTT switch.

A third input for CTCSS tones is also provided.

#### **(\*)2.5.3 KEYING INPUTS**

There are four ways to key the exciter:

1. Pull the Tx-key line low (pin 13 on the D-range 1 [PL100] ) at the rear of the set);
  2. Push the “Carrier” button on the front panel - this will inhibit all audio;
  3. Use the PTT button on the local microphone, which disables audio from the line;
  4. Via the opto-key inputs (pins 11 and 12 on the D-range 1[PL100]) where electrical isolation is required.
- This features a constant current sink (Q270) to ensure reliable activation of the opto-coupler (IC250) at low keying voltages.

#### **2.5.4      COMPRESSOR**

The input signal is fed via a current controlled attenuator (Q230, Q220) to a high gain stage (IC 230) from which the output signal is taken. This signal is passed to a comparator (IC 230) which toggles whenever the audio signal exceeds a DC threshold determined by RV220. Thus, the comparator produces a square wave whose mark-space ratio is determined by the amplitude of the audio signal. This square wave pumps up the reservoir capacitor (C 233) which controls the attenuator (Q230, Q220), thus completing the feedback loop.

The compression level is set by adjustment of the comparator threshold (RV220).

**Note:** Although the high dynamic range of the compressor allows the use of very low audio signal levels, such conditions will be accompanied by a degradation of the signal to noise ratio. Very low audio input levels should therefore be avoided where possible.

#### **2.5.5      OUTPUTS TO MODULATORS**

The output signal from the limiter (IC210, IC230) is added to any incoming CTCSS tone at a summing amplifier (IC 260). The signal is then low pass filtered (IC260) and split to supply the two modulators.

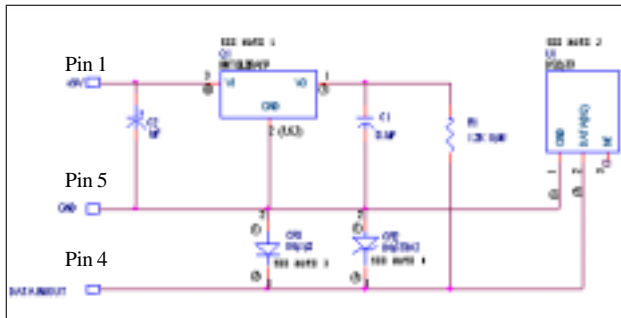
(\*) Since the VCO modulator is a true frequency modulator, its audio is simply buffered (IC 260). The reference modulator, however, is a phase modulator and its audio must first be integrated (IC210).

(\*) It is vital that the audio levels to the modulators are accurately set, relative to each other. Hence the inclusion of level adjustment in the reference modulator path. Once set, adjustments to absolute deviation may be made only by IC220, a 256-step 10k electronic potentiometer (EPOT), which is controlled via PGM800Win. The EPOT is made up of 256 resistive sections (representing approximately 39Ω each) which can be individually

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addressed by the microcontroller. Each section can be switched in or out of circuit to achieve the required total resistance, thus adjusting the absolute deviation level.

### 3. Schematics of the memory module



The other related schematics are supplied along with the production procedure, in Appendix C.