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RST-504 DOCUMENT SUMMARY

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## D. THEORY OF OPERATION

[Information pertaining to audio panels equipped with the RST-506 Intercom package is shown in brackets].

The basic function of the RST-504 is to condition and mix the various audio signals in the aircraft and amplify them for detection in the aircraft speaker and headphones. Also included in the circuitry are a marker beacon audio attenuation circuit [a voice-activated (VOX) intercommunication system], and automatic routing of microphone audio and key signals.

In normal operation, one or more of the eight possible inputs will be amplified and routed to either speaker or headphones, selected by switches S2 through S9. The RST-504 also has a NORMAL / BYPASS switch, S1, which acts as a fail-safe should the panel ever malfunction. In the bypass position, the COM 1 speaker and phones outputs are hardwired to the speaker and phones, providing COM 1 radio operation.

The audio panel operation will be explained in six sections -- General, Speaker Amplifier, Phones Amplifier, Intercom, Marker Attenuation Circuit, and Power Supply. Refer to the chassis Schematic (Drawing #504-7800) and PC050 schematic (Drawing #504-7000) for clarification during circuit descriptions.

### 1. General

- a. Low level (headphone) audio from COM 1, COM 2, ADF, AUX, DME, NAV 1, and NAV 2 enters the audio panel at J1 and J3 and is routed to switches S3 through S9 where it is loaded to ground with a 560 ohm resistor. These switches determine whether the audio is terminated (off), or routed to the speaker or phones amplifier. The MKR BCN audio goes through conditioning circuitry before it is switched to the speaker of phones amplifier.
- b. The AUTO switch routes the COM audio from the radio selected by the transmit switch (XMIT 1/XMIT 2) to either the speaker or phones amplifier as selected. For example, with the AUTO switch in the "speaker" position and XMIT 1 selected, COM 1 audio would be present in the speaker. If the XMIT switch is flipped to the XMIT 2 position, COM 2's audio is provided to the speaker (regardless of the position of either COM 1 or COM 2 switches). The xmit switch, S11 switches the key and audio lines for COM 1 and COM 2 and the COM 1 and COM 2 audio for the AUTO switch.
- c. COM 1 and COM 2 speaker audio signals are terminated to ground through the 10 ohm resistors (R100 and R201).

### 2. Speaker Amplifier

(see Drawing #504-7000.)

- a. Selected low level audio from S2 through S9 enters PC050 at points AS through HS and is summed together at R109 (with the exception of AUX audio from S6, which is introduced later). Operational amplifier U102D attenuates this signal 3 dB with the output appearing at U102-14. Under normal operation, this attenuated audio is applied to the main amplifier input (U101-8) through R111 and C104. R123 sets the input impedance of U101.
- b. A portion of the audio from U102-14 is fed through C103 and R112 to the AUX input muting circuit. The audio is amplified by U102C (10X) and rectified to a dc level by CR101 and C106. This dc level is applied to comparator U102B at the inverting input, pin 6. R117 sets the dc level at the non-inverting input, U102-5. The voltage at C106 is around 0 volts and there is little or no audio at U102-14. This potential rises as the audio level increases. When the level at U102-5 exceeds that at U102-6, the output at U102-7 rises to around 10 volts. This provides the bias voltage required to turn on Q101. Audio from the AUX input (XS) is amplified by Q101 and applied to the amplifier (at U101-8) through C110. When the level at U102-6 exceeds that at U102-5, U102's output drops to around 0 volts turning off Q101 and preventing any AUX audio from being passed to the main amplifier (U101).
- c. The speaker must be muted during both transmit [and ICS functions] to prevent feedback. Q102 and Q103 provide this muting. IN transmit, point KY is pulled low by the key line. This provides the bias required to turn on Q102, bringing its collector voltage up to about 10 volts. This, in turn, provides the bias voltage required to turn on Q103 and shunt any audio present at U101-8 (the main amplifier input) to ground, thus muting all speaker audio. [In the ICS position, 10 volts is applied to point IC which turns on Q107, thus bringing its collector voltage to ground and muting the speaker as described above.] The main speaker amplifier, U101, is a TBA 810 power integrated circuit. Its gain is set at approximately 60 with R128 and C112. The output at U101-12 passes through dc blocking capacitor C117 to OS, the speaker output.

## 3. Phones Amplifier

- a. The phones amplifier circuitry is very similar to that of the speaker amplifier. Low level audio is input at points AP through FP and [R191 with panels equipped with intercom option]. The audio is attenuated by U105C and applied to the phones amplifier at U106-2. The AUX muting circuit involving U105B, U105A and Q104 is similar to that described in the speaker amplifier section, above.
- b. The COM 1 and COM 2 inputs are conditioned so that when one COM is transmitting, the other one is muted (i.e. COM 2 is muted when transmitting on COM 1). This prevents possible bleedover between radios. COM 1 audio is input at point HP and applied to U104-2 through R158 and C131. As long as point OL (COM 2 key line) is at around 12 volts (unkeyed state), audio will be amplified by U104A (unity gain) and summed with the rest of the phones inputs at U105-9. If OL is near 0 volts (COM 2 keyed), then biasing is removed from U104-3 and little or no audio will be present at

U104-1. The above description also applies to the COM 2 circuit incorporating GP, IL, and U104B.

c. The main phones amplifier is a LM386 low voltage audio power amplifier. The gain is set at approximately 60 by R162 and C134. The output at pin 5 is set internally at V+/2 (5 volts for a 10 volt supply) to provide maximum output swing. After passing through dc blocking capacitor C137, headphone audio is output at point OP.

## 4. [Intercommunications System]

- a. [Built into the RST-504 audio panel is an optional voice-activated intercom system. The system is activated when S12 is rotated to the "ICS ON" position by providing a 10 volt dc signal to point IC which activates the ICS circuitry. The 10 volt signal at point IC turns on Q106 and Q107. The collector of Q107 is pulled to ground muting the speaker amplifier through R126 and turns on the MM marker lamp through CR111 and R175) one-half brilliance to indicate that the ICS is activated.
- b. The emitter voltage of Q106 rises to approximately 9.3 volts to bias "ON" the four microphones through R176 (PM), R177 (CM), R178 (TM), and R179 (FM). These four microphones' signals are then summed at C143 and amplified (3X) by U103B. Audio from U103-7 passes through R187 and C148 to ICS volume control R191. A percentage (0-100%) of this audio is selected from the wiper of R191 for presentation to the phones amplifier, U106, via the summing amp U105C.
- c. To provide voice activation, a squelch circuit is needed. A portion of the audio from U103-7 is rectified by CR112 and C146 to a dc level, and applied to comparator U103A, pin 2. When the level set by R192 (ICS SQUELCH) is greater than that at pin 2, the output of the comparator goes to near 10 volts and turns on Q108 through CR113 and R190. This shunts any audio present at C148 to ground. When there is sufficient audio to "break" the squelch, Q108 is turned off and audio is routed to the headphone summing amplifier, U105C.]

### 5. Marker Attenuation Circuit

- a. Once a preset marker audio level has been achieved, the marker attenuation circuit will attenuate the marker audio to a comfortable listening level. After the aircraft has passed over the marker beacon, the circuit resets itself to maximum sensitivity.
- b. Marker beacon audio is fed into point MI on PC050 and loaded to ground with R163. The signal is amplified (2X) by U104C and output at pin 8. In normal, low level operation, this audio will be present at the marker beacon audio output (MO) and applied to S2 (MKR BCN) for selection to the speaker or phones amplifier.

c. To activate the attenuation circuit, a portion of the audio from U104-8 is rectified by CR110 and C140 to a dc level and applied to comparator input U104-12. R170 (MKR TRIG ADJUST) sets the level at U104-13. As the audio level at U104-8 increases and the level at U104-12 exceeds that at U104-13, the output of U104-D rises to about 10 volts and turns on Q105, bringing its collector voltage to ground. A portion of the audio passing through R166 is then shunted to ground through R172. The degree of attenuation is adjustable from about 3 dB to over 60 dB by R172.

# 6. Power Supply

- a. Power for the RST-504 comes in to PC050 at point VP. The 12-15 volt input is filtered by C149 and C150. The positive supply for the marker/indicator lamps is fused with R193 (a 10 ohm, 1/4 watt resistor) and output at point ML.
- b. CR114, C151, and R194 provide the stable 10 volt reference for Q109, the series pass element. The fairly high current 9.3 volt supply is filtered by C152. To obtain the V+/2 supply, R197 and R198 form a voltage divider to produce the V+/2 signal at U105-12. U105D is hooked up in a follower configuration to provide the required current drive for the V+/2 (5 volt) supply.



Section G



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Schematics and Drawings

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Schematics and Drawings



