

CONTROLS:

From left to right:

UPPER POSITION: Magneto-dynamic pick-up

LOWER POSITION:

Crystal pick-up

CONSUMPTION AND LINE VOLTAGE: 85 W (117 V-60 cycle).

HUM LEVEL: -60 db with respect to 6 W + 6 W output.

3. Volume control Treble control

POWER OUTPUT: Measured across a load resistance of 800+800 Ω. Selector switch

BASS CHANNEL: 6 Watts at 6% distortion (frequency = 90 c/s) The input is at the rear of the unit switch (SK4).

# A FEW DETAILS ABOUT THE CIRCUIT DIAGRAM:

TREBLE CHANNEL: 6 Watts at 6% distortion (frequency = 4000 c/s).

The AG9008 is a quality amplifier which amongst other things is suitable for operation with:-

- 1. Crystal or magneto-dynamic pick-up.
- 2. Tuner or magnetophone.

ON-OFF switch

Bass control

The amplifier is equipped with a continuously variable bass and treble control. A switch (SK3) is incorporated for adaptation to 3 recording characteristics.

The amplifier is designed according to the bi-ampli principle.

The output is derived from two single ended push-pull output stages utilizing 2 x 6CW5 tubes for each channel.

The pre-amplifier which drives both output stages via the cross-over filter, employs a 16A8 tube and both bass and treble controls have been incorporated in this circuit.

The cross-over filter has its cross-over point at 450 c/s. (The frequency whereby equal power is developed in each channel).

The input to the (control) driver stage is made up of a double tapped 500 KΩ volume control (R48-R48a-R48b). At low positions of the volume control the high notes as well as the low notes are boosted a little, so that better matching is obtained.

The input to the volume control is applied by the correction switch (SK3) either to the tuner input or via filters to the output of the pick-up correcting pre-amplifier (12AX7).

These filters which consist of R.C. elements compensate for the recording characteristics which differ a few db's from each other.

The correcting pre-amplifier B1 (12AX7) is switchable for magneto-dynamic or crystal pick-up (switch at the rear side of the chassis).

The switch is arranged so that the output voltage is the same for both pick-up types.

The correction for the output voltage of the pick-up as a function of the frequency is such that for the various recording characteristics only a certain attenuation of definite frequency ranges is necessary via the above mentioned filters, in other words, in the circuit around Bl and Bl' the maximum required bass and treble of all recording characteristics is corrected.

# HUM AND RATTLE:

As selenium cells, which are of very low internal resistance, are used in the rectifying circuit, strong current pulses arise in the supply transformer. These current pulses appear as small voltage pulses in the heater circuit.

Pin 5 of the 12AX7 transfers these pulses to pin 7 (grid B1') via the tube socket capacity, so that a rattle is audible in the treble channel. By coupling the pulses of pin 4 (being in phase opposition) to the pulse of pin 7, via the wiring capacity, this rattle can be removed entirely.

This can be done as follows (if rattle causes trouble).

The filament lead of pin 4 is bent a little towards pin 7, or is pushed a little away from same, so that the rattle becomes nil.

# THE SELECTOR SWITCH:

# Position 1:

In this position the tuner input is connected through DC blocking condenser C15 (.022 µfd) to the top of the volume control. C15 is necessary because tuners often employ a cathode follower circuit which has a leakage current of a few uA.

The output of the pre-amplifier (12AX7) is shortcircuited across R51, C46 and C27 the junction point being grounded through SK3. Although not a 100% shortcircuit, this is done, because otherwise B1 and B1' could start oscillating at a very low frequency (3-8 c/s) due to cathodes being tied together.

Position 2: (see fig. 3)

In this position the pick-up pre-amplifier unit is connected through C46, R51 to the top of the volume control.

C46 and R51 match the output voltage of the pre-amplifier to the recording characteristic. The input of the tuner is grounded in this position of SK3.

Position 3: (see fig. 4)

C27 is connected in series with C46, R51 and at the same time C38 is connected in parallel with C46, R51.

C27 gives an attenuation of the bass range, C38 boosts the treble range.

The input of the tuner is shortcircuited via C15 in this position.

Position 4: (see fig. 5)

C27 is shortcircuited, so that the low notes are less attenuated; at the same time C38 is replaced by C48 connected in parallel with C46, R51. This boosts the treble range.

The tuner input is grounded via C15.

# BASS-CONTROL:

Part of the plate A.C. voltage of B2' is taken from R28 and applied to the grid via C22, C20 and R47.

C22 and C20 have a low impedance for high frequencies, so that the negative feedback for the high frequencies is independent of the position of R47.

TUBES: B1 : 12AX7 B2 : 16A8 B3 : 6CW5 6CW5 **B4** 

B5 :

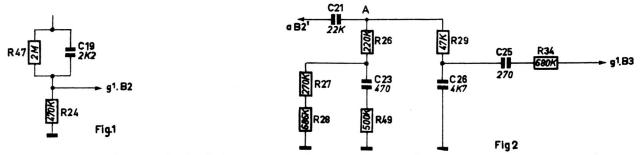
C22 and C20 can be overbridged more or less with R47, this has no influence on the high frequencies of the negative feedback signal, but it has for the low ones.

C22 and C20 have a large impedance for low frequencies. If control R47 is in the upper position, the low frequencies receive only a small amount of negative feedback. The result of this is that the low notes are favoured above the rest of the frequency spectrum. When R47 is moved downwards, the negative feedback for the low notes becomes stronger and stronger, so that the low notes become weaker.

If R47 is in the lower position, C22 and C20 are short-circuited, so that all frequencies now obtain an equally strong feedback.

In the lower position of R47 all frequencies receive equally strong feedback, so that the output signal of B2' should be straight.

This is not the case, however, since C19 operates as coupling capacitor in this position.



This can be understood as follows: C19 and R47 which are connected in parallel, together form a much greater impedance for low frequencies than R24. (see fig. 1). From this it follows that the low frequencies are attenuated with respect to the high ones.

When R47 is rotated upwards again, the influence of C19 becomes smaller and smaller, so that the low notes are less attenuated.

The influence of the filter C22-R25, C20-R24 becomes greater and greater, as a result of which the high frequencies obtained a stronger feedback, whereby the low notes are favoured.

#### TREBLE CONTROL:

R49 is the treble control. C23 and R49 (wholly or partially) are connected in paralled with R27 and R28.

For low notes this parallel connection has no influence due to the small capacitance of C23.

B4 and B3 form the output stage for high notes. The necessary control signal is applied to the control grid of B3.

If R49 is in the lower position, we get the situation shown in Fig. 2.

Only the parts which are of importance for this treble control, have been indicated.

We see that the plate A.C. voltage of B2' is between point A and earth. via a parallel and series connection of resistors and capacitors. R29 and C26 form a voltage divider. As the impedance of C26 is much smaller for high notes than the impedance of R29, the high notes will be attenuated. So B3 receives few high notes. As R49 is rotated clockwise, C26 is moved further from ground and part of R49 is connected between ground and C26. The high notes are therefore attenuated less and less. When R49 is in the upper position, the high notes are no longer attenuated.

C23 is now connected in paralled with R27-R28. C23 has a small impedance for high notes with respect to R27-R28.

The negative feedback voltage across R28 is therefore small for high frequencies, in other words, the negative feedback voltage decreases with increase in frequency. The high notes are therefore favoured. The negative feedback voltage reaches the grid of B2' via C22, C20. As R49 goes downwards, a resistor is placed in series with C23, as a result of which the impedance for high frequencies becomes greater and this means that the high frequencies obtain more negative feedback, as a result of which they are less favoured.

# THE BI-AMPLI PRINCIPLE:

Here the high and low notes for the output stage are divided, whereby the high notes and low notes each have their own output stage and loudspeaker(s).

At point B in the circuit diagram all frequencies are still present. B3 is controlled via the filter C25-R34-R31-R35-R32-C29.

The above filter attenuates the low notes very strongly, so that B3 and B4 receive only high notes.

B5 is controlled via R33-C28 and R18-R61 and R38. The combination R33-C28 attenuates the high notes very strongly, B5 and B6 therefore only receive the low notes.

B5 and B6 form, just like B3 and B4, a so-called single ended push-pull circuit, the working of which will be explained later on.

B3 and B4 obtained negative feedback via R59 in the grid circuit of B3, while B5 and B6 obtain negative feedback via R60-C39 in the grid circuit, and this reduces distortion.

# SINGLE ENDED PUSH-PULL CIRCUIT:

As B3 and B4 operate in the same manner as B5 and B6, the working of B3 and B4 only, will be discussed here (see fig. 6).

Only the most essential components have been shown.

B4 and B3 are connected in series. If no signal is applied to B3, the plate currents are therefore of equal value.

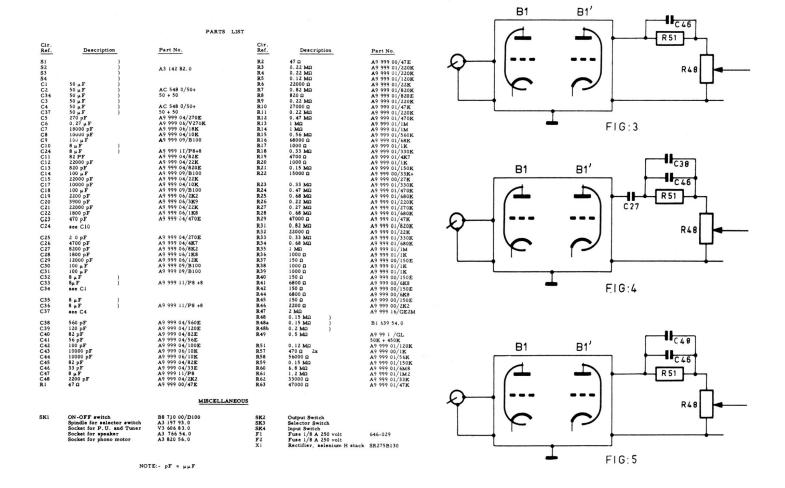
The cathode resistors are also equal, so that the tubes have the same bias.

If we suppose that the grid of B3 obtains a negative pulse at a certain moment, the tube current of B3 decreases. The voltage across R42 therefore becomes smaller.

From this it follows that the grid voltage of B4 also becomes smaller and the plate current of B4 therefore becomes greater.

Via C33 and the loudspeaker a current will flow which is equal to the difference of the momentary values of Ia4 - Ia3. (Ia4 = plate current of B4) (Ia3 = plate current of B3).

The above follows from the first law of Kirchhoff, which states: "The algebraic sum of the instantaneous values of all currents at a definite point (here point S) is zero at every instant".



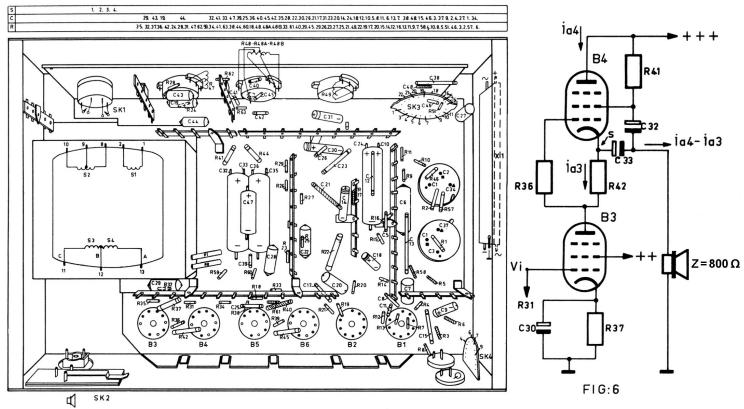


Fig.8