Copyright Warning & Restrictions

The copyright law of the United States (Title 17, United States Code) governs the making of photocopies or other reproductions of copyrighted material.

Under certain conditions specified in the law, libraries and archives are authorized to furnish a photocopy or other reproduction. One of these specified conditions is that the photocopy or reproduction is not to be "used for any purpose other than private study, scholarship, or research." If a, user makes a request for, or later uses, a photocopy or reproduction for purposes in excess of "fair use" that user may be liable for copyright infringement,

This institution reserves the right to refuse to accept a copying order if, in its judgment, fulfillment of the order would involve violation of copyright law.

Please Note: The author retains the copyright while the New Jersey Institute of Technology reserves the right to distribute this thesis or dissertation

Printing note: If you do not wish to print this page, then select "Pages from: first page # to: last page #" on the print dialog screen



The Van Houten library has removed some of the personal information and all signatures from the approval page and biographical sketches of theses and dissertations in order to protect the identity of NJIT graduates and faculty.

ABSTRACT

Title of Thesis: Simulation of Voice and Data Integration Urmila S. Patel, Master of Science, 1991 Thesis directed by: Professor Irving Wang

A simulation model of integrated voice and data is developed. The data packets are transmitted during voice inactivity (i.e. silence duration). A statistical model of voice (talkspurt) and silence duration is generated. Inter-arrival time between data packets is modeled using geometrical distribution. The data traffic statistics is simulated for a given voice activity model and various data packet arrival rates. Four important results are obtained by performing this simulation: 1. Average data packet delay versus arrival rate. 2. The optimal ratio of data packet arrival rate to average data packet delay. 3. Probability of wait time for various data packet arrival rates. 4. Probability of zero waiting time versus arrival rate. These results can be utilized to optimally select the data traffic intensity. Several plots have been generated to provide more insight of the voice and data integration.

Simulation of Voice and Data Integration

.

1

by

URMILA S. PATEL

Thesis submitted to the Faculty of the Graduate school of the New Jersey Institute Of Technology in partial fulfillment of the requirement for the degree of Master of Science in Electrical Engineering

May 1991

APPROVAL SHEET

Title of Thesis:	Simulation of Voice and Data Integration
Name of Candidate:	Urmila S. Patel
	Master of Science in Electrical Engineering, 1991
Thesis Approved:	

Dr. Irving Wang Assistant Professor Electrical Engineering Date

Dr. Edwin Hou Assistant Professor Electrical Engineering

Date

Dr. Meng-Chu Zhon Assistant Professor Electrical Engineering

Date

VITA

Name : Urmila S. Patel

Degree and Date to be Conferred : M.S.E.E., 1991.

Secondary Education : M.R.V. High School, 1983.

Collegiate Institutions Attended	Date	Degree	
New Jersey Institute of Technology	09/88-05/91	M.S.E.E.	
New Jersey Institute of Technology	09/8405/88	B.S.E.E.	

Major : Electrical Engineering

Minor : Computer Networks

To my husband Kartik and brother Kumud -

TABLE OF CONTENTS

Chapte	er	Pages
1.	INTRODUCTION	1
2.	PROTOCOL AND MODEL DESCRIPTION	4
	2.1 System Model	4
	2.2 Protocol Description	9
3.	SIMULATION DESIGN AND EXPERIMENTS	12
	3.1 Voice Process Generation	12
	3.2 Voice Inactivity Process Generation	15
	3.3 Data Packet Arrival Process Generation	16
	3.4 Voice/Data Integration Statistics Generation	16
4.	SIMULATION RESULTS	22
	4.1 Voice Simulation Result	22
	4.2 Voice Inactivity Simulation Result	22
	4.3 Data Packet Arrival Simulation Results	23
	4.4 Voice/Data Integration Simulation Results	32
	4.5 Average Data packet delay Verus Arrival rate	32
	4.6 Ratio of Average arrival rate to Average delay	33
	4.7 Probability of Wait time Versus Arrival rate	38
	4.8 Probability of Zero wait time Versus Arrival rate	38
5.	CONCLUSION	46
APPEN	NDIX A. SOURCE CODES	47

APPENDIX	В.	DERIVATION	OF	EXPECTED	VALUE	*****	62
SELECTED	BIB	LIOGRAPHY .				•••••	65

.

LIST OF TABLES

Table		Pages
4.1	AVERAGE DATA PACKET DELAY	34
4.2	OPTIMUM ARRIVAL RATE	36
4.3	PROBABILITY OF ZERO WAIT TIME	44

•

LIST OF FIGURES

Figures		Pages
2.1	INTEGRATION OF VOICE AND DATA TRAFFIC	5
2.2	SPEECH ACTIVITY DETECTOR	6
2.3	VOICE/DATA INTEGRATION PROTOCOL	10
3.1	VOICE/DATA INTEGRATION FLOWCHART	18
4.1	VOICE STATISTICS	24
4.2	SILENCE DURATION STATISTICS	26
4.3	DATA PACKET ARRIVAL STATISTICS AT RATE 4	28
4.4	DATA PACKET ARRIVAL STATISTICS AT RATE 8	29
4.5	DATA PACKET ARRIVAL STATISTICS AT RATE 12	30
4.6	DATA PACKET ARRIVAL STATISTICS AT RATE 16	31
4.7	VOICE/DATA INTEGRATION STATISTICS	35
4.8	OPTIMUM ARRIVAL RATE	37
4.9	PROBABILITY OF WAIT TIME AT RATE 4	40
4.10	PROBABILITY OF WAIT TIME AT RATE 8	41
4.11	PROBABILITY OF WAIT TIME AT RATE 12	42
4.12	PROBABILITY OF WAIT TIME AT RATE 16	43
4.13	PROBABILITY OF ZERO WAIT TIME VS.	
	ARRIVAL RATE	45

v

CHAPTER 1

INTRODUCTION

In the last ten years, there has been a great deal of activity in integrated telecommunication networks. Public telephone and telecommunications networks are rapidly evolving from analog technology to the exclusive use of digital technology. The Integrated Digital Network (IDN) will combine the extensive geographic coverage of the telephone network with the data carrying capacity of digital data networks in a structure called the Integrated Service Digital Network (ISDN).

ISDN will become a worldwide public telecommunications network that will service a wide variety of user needs. ISDN will support a variety of services related to voice communications (telephone calls) and nonvoice communications (digital data). These services are to be provided in conformance with standards (CCITT recommendations) that specify a small number of interfaces and data transmission facilities.

Many different types of services will be available from this system, such as an ISDN data transmission service that will allow users to connect their ISDN terminal or computer to another one anywhere in the world. A new communication service called Videotex is expected to become widespread with ISDN. Videotex provides an interactive access to a remote database by a person at a distant terminal. Simultaneous voice and data transmission is another application being considered for ISDN.

The movement toward digital technology has been pushed by the competitive desire to lower cost and improve quality of voice transmission and networking services. In an analog network, signals may pass through one or more intermediate switching centers before reaching the destination. At each switching center, the incoming FDM carrier has to be demultiplexed and demodulated by an FDM channel bank before being switched by a space-division switch. After switching, the signals have to be multiplexed and modulated again to be transmitted. This repeated process results in an accumulation of noise, as well as high cost. Whereas, in a digital network, incoming voice signals are digitized using PCM and multiplexed using TDM, so Time-division digital switches along the way can switch the individual signals without decoding. Furthermore, separate multiplex/demultiplex channel banks are not needed at the intermediate offices, since that function is incorporated into the switching system.

A number of small-scale ISDN facilities have been developed, but the promised extension of ISDN to encompass worldwide public telecommunication is not yet a reality. Also, digital service has not yet been extended to the end user. Telephones are still sending analog data to the end office where they must be digitized. Lower-speed (<56 kbps) end-user digital service is commonly available via leased lines at present, and higher-speed leased services are being introduced.

This thesis describes and analyzes transmission of data packets (nonvoice) during voice communication.

The main objectives of this simulation project are:

- 1. To obtain Voice and Data statistics, for use in simulation program.
- 2. To obtain data packet delay statistics.
- 3. To achieve or to select the optimal ratio of data packet arrival rate to average time delay per packet.
- 4. To obtain waiting statistics for data packets.
- 5. To provide a general scheme for integrated voice and data modeling which can be used in mobile voice/data communication or CATV networks.

Chapter 2 provides protocol and model description. Chapter 3 gives a detailed description and analysis of generating talkspurt and silence duration used in simulation design. Simulation results are discussed in chapter 4.

CHAPTER 2

PROTOCOL AND MODEL DESCRIPTION

This chapter presents the description of a model on which voice/data integration protocol is tested. It also provides an overview of protocol and defines the assumptions used to develop the protocol.

2.1 SYSTEM MODEL:

2.1.2 MODEL DESCRIPTION:

The system considered here has one voice source and one data packet source sharing one voice channel for transmission purpose. As shown in figure 2.1 two types of traffic arrive at a transmission link. The voice traffic is digitized. It can be PCM, ADPCM or any other compressed voice. A voice detection unit may be embedded into the controller. A controller is used to schedule the traffic on arrival. The controller would know each user's frame bandwidth requirement and would decide whether to immediately transmit an arriving data packet or schedule it for later transmission. Figure 2.1 conceptually explains how voice and data integration is modelled.

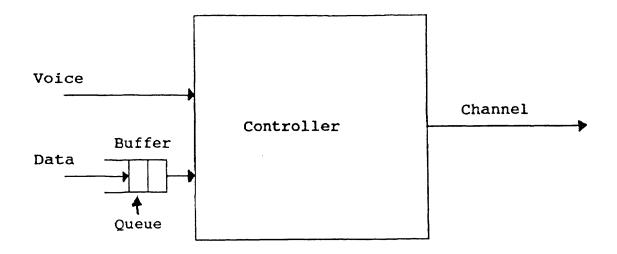


Figure 2.1 Integration of Voice and Data Traffic

Data packets arrive at rate λ packets/sec with inter-arrival time geometrically distributed. During voice inactivity, data packets are placed for transmission. The key to combining voice and data onto a common channel is the concept that speech can be modelled as short bursts of vocal energy (talkspurt) separated by gaps contain no vocal energy (silence duration). Thus, voice detection is a crucial part of the system since it determines when to start and stop transmitting data packets.

The model described here is a representation of (1) Collocated voice and data sources using a single channel and (2) Two stations, one transmitting voice and the other transmitting data through a common node in a network using the same channel.

A. Voice Model:

Speech first passes through a speech detector which generates an on-off speech pattern. Figure 2.2 depicts a functional block diagram of the speech detector. Incoming digital speech signals are used to compute speech energy (amplitude). A threshold detector is set at this point to detect the presence of an amplitude above some fixed value. The threshold detector triggers a flip-flop between clock pulses. A pulse will be output synchronously with a frame clock. This pulse indicates that speech energy crossed the threshold, which indicates an "on interval." The absence of a pulse indicates an "off interval." A spurt is an unbroken sequence of on intervals. A gap is an unbroken sequence of off intervals. Using this method, speech can be transformed into a spurt gap pattern.

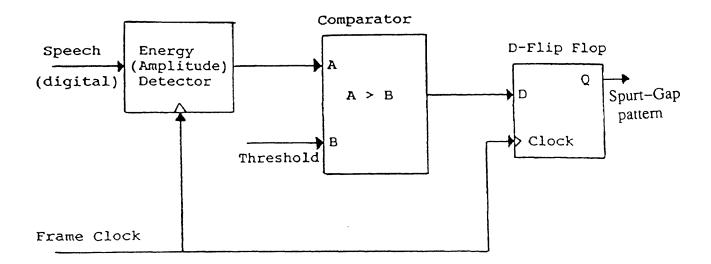


Figure 2.2 Speech Activity Detector.

Talkspurt:

Speech waveform is divided into fixed-size frames. This is accomplished by use of a frame clock to sample the speech detector. It ensures that on/off intervals will be an integral multiple of frame lengths. The pattern obtained corresponds to talkspurt and silence duration. Talkspurt is a time period that contains a sequence of speech sounds unbroken by pauses [1]. Each frame is evaluated for energy content, and if total energy content falls above a certain threshold, the frame is taken as a talkspurt. Here we consider 1 frame to be 22.5 milliseconds. The length of the talkspurt is described by a geometric distribution.

Silence Period:

A silence period can occur between syllables and between words, as well as between phrases and sentences. Silence gap detection is accomplished by dividing the speech waveform into fixed-size frames. Each frame is evaluated for energy content, and if total energy content falls below a certain threshold, the frame is taken as a silence. Here we consider 1 frame to be 22.5 milliseconds.

B. Data Model:

Poisson arrival is most often used to characterize random arrival of the data packets at the node. It can also be expressed by exponential inter-arrival time. The system model considered here is discrete. The discrete geometric distribution is the closest fit to the exponential distribution. Thus the geometric distribution for inter-arrival time of data packets is considered. The data packet length is 1 frame (22.5 milliseconds). The inter-arrival is an integral multiple of frame periods.

7

Longer data packets (n frames long) can be represented as n packets with an interarrival time of 1 frame.

2.1.2 ASSUMPTIONS:

- * All variables are measured using a frame as an unit.
- * A frame clock is the heart of the simulation.
- * A frame length is 22.5 millisecond long.
- * The voice talkspurt length (in terms of frame length) is geometrically distributed with average length of 7.54 frames [4].
- * Talkspurt is an integral multiple of frame length.
- * The silence duration (in terms of frame length) has weighted geometric distribution [4]. An average silence duration is 5.95 frames long.
- * Silence is an integral multiple of frame lengths.
- * An inter-arrival time between two data packets is assumed to be geometrically distributed. The parameter RHO is equal to
 1.0 (Ar. rate * .0225).
- * All data packets are one frame long.
- * The detector/controller is assumed to be ideal such that the voice detection is instantaneous.
- * Buffer capacity for data packets is infinite.
- * Probability of having zero inter-arrival time (simultaneous arrival) is zero.

- * The length of one talkspurt and silence duration is independent of another talkspurt or silence duration.
- * Talkspurt and silence duration are always alternated.
- * Average talkspurt or silence rate is 1/(Ave. talkspurt time + Ave.
 silence duration) = 1/.0225*(7.54+5.95) = 3.29 /sec.
- * Channel capacity is fixed C bits/second.
- * The service time of data packets depends on length of the packet, since data packet length is one frame, the service time is 1 frame period long.

2.2 **PROTOCOL** DESCRIPTION:

A protocol for data transmission over a digital voice channel is explained here. Data packets arrives randomly and inter-arrival time follows geometric distribution. When voice is active (during talkspurt), all data packets are stored in the queue. After that talkspurt silence duration follows. During the silence period, the packets are transmitted. The packets are served on First Come First Serve (FCFS) basis. During the data packet transmission period, if any more packets arrive, they are placed in the queue and remain there until service is available. The service availability depends on average silence duration as well as data packet arrival rate. If during the silence period, there is nothing in the queue and a data packet arrives, it is transmitted immediately.

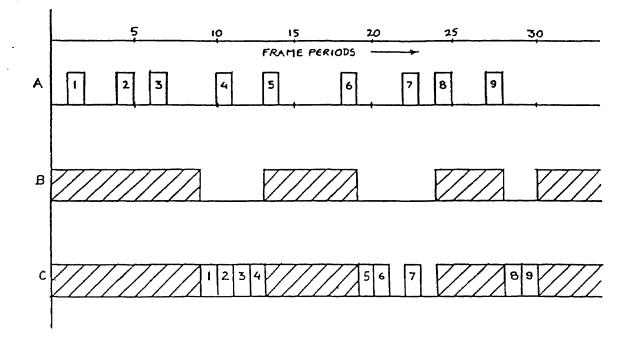


Figure 2.3 Voice/Data Integration Protocol A) Arriving data packet B) Voice activity C) Integration of voice and data.

Figure 2.3 above depicts the voice/data integration protocol. The first talkspurt is 9 frames long. That is followed by silence duration of 4 frames. The next talkspurt is 6 frames long followed by silence duration of 5 frames. The third talkspurt is 4 frames long followed by silence duration 2 frames long. A total of 9 data packets arrived during this period. They arrived at the end of 1^{st} , 4^{th} , 6^{th} , 10^{th} , 13^{th} , 18^{th} , 22^{nd} , 24^{th} and 27^{th} frame. Data packets numbered 1, 2 and 3

arrived during a talkspurt so they were placed in the queue. At the end of the first talkspurt (9th frame), the transmission of the queued packets began. The 1st, 2nd, and 3rd packets were transmitted during frames 10, 11 and 12 respectively. During transmission of these three packets, another packet arrived. Since there was one frame worth of silence duration available, the 4th packet was transmitted during the 13th frame. The fifth data packet arrived at the beginning of the next talkspurt and sixth one during the talkspurt. Both were placed in the queue until a silence period was available at the end of the 18th frame. Both packets were transmitted during the 19th and 20th frame leaving an empty queue. The 7th data packet came in at the beginning of the 22nd frame. It was transmitted immediately since the queue was empty. The eighth data packet arrived at the beginning of third talkspurt and ninth one during the talkspurt. Both were placed in the queue until a silence period was available at the end of the 27th frame. Both packets were transmitted during the 28th and 29th frame. Both were placed in the queue until a silence period was available at the end of the 27th frame. Both packets were transmitted during the 28th and 29th frame.

CHAPTER 3

SIMULATION DESIGN AND EXPERIMENTS

This chapter discusses the simulation design used in the experiment performed. FORTRAN is used for the implementation of the model and simulation. Simulation codes are given in Appendix A. Three different process models were developed: 1. Speech activity or talkspurt length. 2. Silence duration and 3. Data packet interarrival time. The voice/data integration process model is generated using a protocol in conjunction with three process models.

3.1 VOICE (TALK SPURT) PROCESS GENERATION:

3.1.1 ALGORITHM

To obtain statistical data for a voice (talkspurt), an algorithm to generate random numbers with a geometric probability distribution function was developed. Theoretical values were also calculated. Collected results are: average talkspurt length in frames, and probability distribution curves for simulated and calculated values. Derivation of talkspurt:

The measured talkspurt length can be modelled by a geometric distribution with a probability density:

$$f_x(k) = (1-r) r^{k-1}$$
 $k = 1,2,3,4....(1)$

Where $f_x(k)$ is the probability that the talkspurt is k frame long and r=0.8674[4].

Random numbers of the geometric variable G:p can be computed from the set of uniformly distributed random numbers U. Before we proceed to simulate the geometric random variable, we will first look at the general algorithm for simulating X [2]. X is a random variable that takes the values: 1,2,3,... and with

$$p_i = P_r(X=i)$$
 for $i \ge 0$.

Select a uniform random variable U(0,1).

Set X=0 if
$$0 \le U < p_0$$
, and
Set X=j if $\sum_{i=0}^{j-1} \sum_{i=0}^{j} U < \sum_{i=0}^{j} p_i$ for $j \ge 1$

Now it is easy to see that this algorithm works by selecting a value U and observing in which probability interval U lands.

e.g take j=3

Now check
$$\begin{array}{ccc} 2 & 3\\ \Sigma p_i \leq U < \Sigma p_i\\ i=0 & i=0 \end{array}$$

If U lands in this range than set X=3.

「「「「「「「「「」」」」

Now based on this algorithm we will go to simulate the geometric random variable X, which is one of the keys to generating a talkspurt.

$$p_i = P_r(X=i) = r (1-r)^{i-1}$$
 for $i \ge 1$, 0

To operate the above algorithm we need successive cumulative sums of the $\{p_i\}$. Because p_i is of a simple geometric form, cumulative sums are also of a simple form. Here,

13

Thus the algorithm becomes:

Set X=j if
$$1-r^{j-1} \le U < 1-r^{j}$$
 for $j \ge 1$
 $-r^{j-1} \le U-1 < -r^{j}$
 $r^{j-1} \ge (1-U) > r^{j}$ (2)

It is intuitively clear that if U is a U (0,1) random variable, then so is (1-U), Hence we can reduce the labor of arithmetic slightly.

$$\mathbf{r}^{\mathbf{j}-1} \ge \mathbf{U} > \mathbf{r}^{\mathbf{j}} \tag{3}$$

The random variable X resulting from using (2) will have the same geometric distribution as the random variable resulting from using (3).

We set $X = j \ge 1$ if, and only if

 $(j-1) \log_e(r) \ge \log_e U > j \log_e(r)$

We have X = j if

$$(j-1) <= \frac{\log_e U}{\log_e(r)} < j \text{ for } j >= 1 ...(4)$$

Finally, we note that we can express (3) very simply by setting

$$X = 1 + \frac{\log_e U}{\log_e(r)}$$
(5)

Random variable X will take an integer value of the expression in eq.(5). This geometrically random variable X is used to generate the length of talkspurt. The average talkspurt duration is 7.541 frames or 169.7 milliseconds. An expression for expected value of talkspurt duration is derived in appendix B equation (6).

3.1.2 PROGRAM

The FORTRAN source code is given in appendix A.1.

3.2 VOICE INACTIVITY (SILENCE DURATION) PROCESS GENERATION:3.2.1 ALGORITHM

Weighted geometric distribution is used to generate the silence duration statistics. The user inputs the seed value and the amount of total numbers to be generated. Using the geometric probability density function, a theoretical value is calculated. Calculated and simulated probability distribution curves, and average silence duration length are collected

Derivation of Silence Duration:

Silence duration length can be modeled as the truncated sum of two suitable weighted geometric distributed probability density function [4]:

$$f_{I}(k) = C_{1}(1-r_{1})r_{1}^{k-1} + C_{2}(1-r_{2})r_{2}^{k-1}$$
 $k = 1,2,3,....$ frames(6)

As we have seen in a previous derivation, the successive cumulative sum of $\{p_i\} = 1-r^j$ for $j \ge 1$

therefore,

.. **.**

$$\sum_{k=1}^{j} f_{i}(k) = C1 (1-r_{1}^{j-1}) + C2 (1-r_{2}^{j})$$

Set X=j if $C_1 (1-r_1^{j-1}) + C_2 (1-r_2^{j-1}) \le U < C_1 (1-r_1^{j}) + C_2 (1-r_2^{j})....(7)$

Now, by generating the uniform random number U, and observing which probability interval U lands in, the length of the silence duration is generated. The average silence duration is 6.659 frames or 149.8 milliseconds. Refer to appendix B for Expected value derivation.

3.2.2 PROGRAM

The FORTRAN source code is given in appendix A.2.

3.3 DATA PACKET ARRIVAL PROCESS GENERATION:

3.3.1 ALGORITHM

Geometric probability distribution is used to find inter-arrival time. The arrival rate is a parameter. An algorithm is developed to generate geometrically distributed random numbers using equation (5). It computes theoretical as well as simulated probability distribution for four different arrival rates (4,8,12,16).

3.3.3 PROGRAM

The FORTRAN source code is given in appendix A.3.

3.4 VOICE/DATA INTEGRATION STATISTICS GENERATION:

3.4.1 FLOWCHART

The voice data integration simulation flowchart is given in figure 3.1(a), 3.1(b) and 3.1(c).

List of variable used in flowchart and simulation.

IS : Seed value to generate Uniform Random Number

N: Total number of data packets generated (counter)

Q: Total number of data packets in Queue

TV : Talkspurt length

TS : Silence duration

TT : Total TV + TS

TQS : TV + Q

TTV : TT + TV

TA(n) : Arrival time of nth data packets (in terms of frame number)

TD(n) : Departure time of nth data packets (in terms of frame number)

V : Counter for total number of talkspurt / silence duration.

TMAX : Maximum simulation time

LAMBDA : Data packets arrival rate

3.4.2 ALGORITHM

This simulation program provides flexibility to choose the desired value for total simulation time and data packet arrival rate. It prompts for seed value simulation time in seconds and data packet arrival rate in packets/second. Actual simulation time is expressed as an integer multiple of 10000 frames, i.e. 10000 seconds is equals to 444,444 frames. The actual simulation time will be 450,000 frames. Data packets are generated by calling the subroutine Arrival(). The total arrival time is the sum of inter-arrival. The program stops and calculates the final results only if total arrival time is greater than or equal to maximum simulation time. The subroutines Voice() and Silence() are called to generate talkspurt and silence

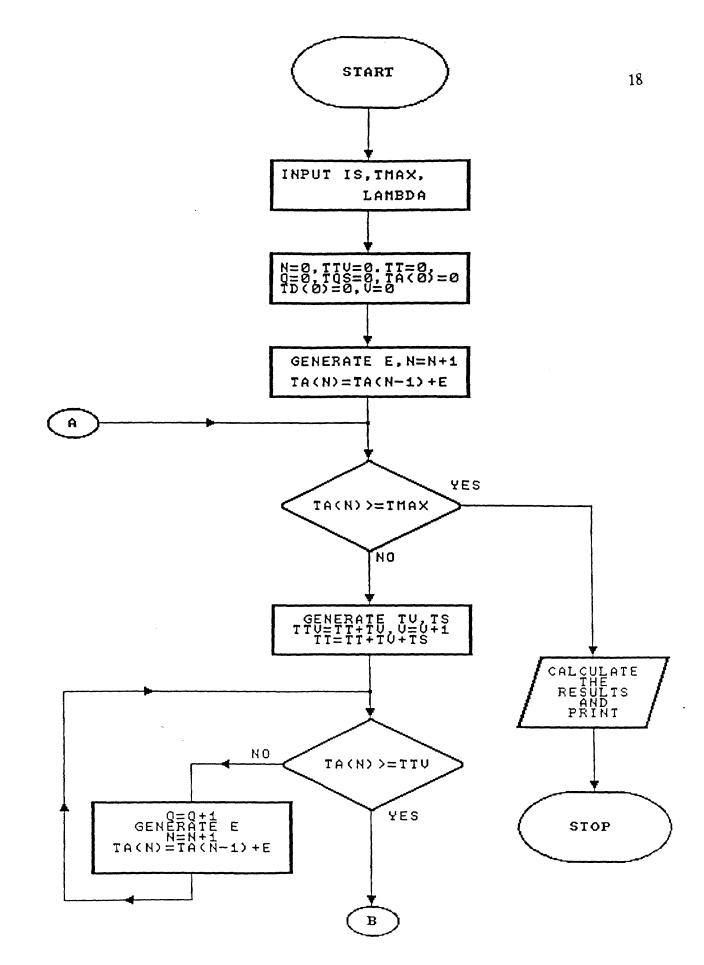


Figure 3.1(a) Voice/Data Integration Flowchart

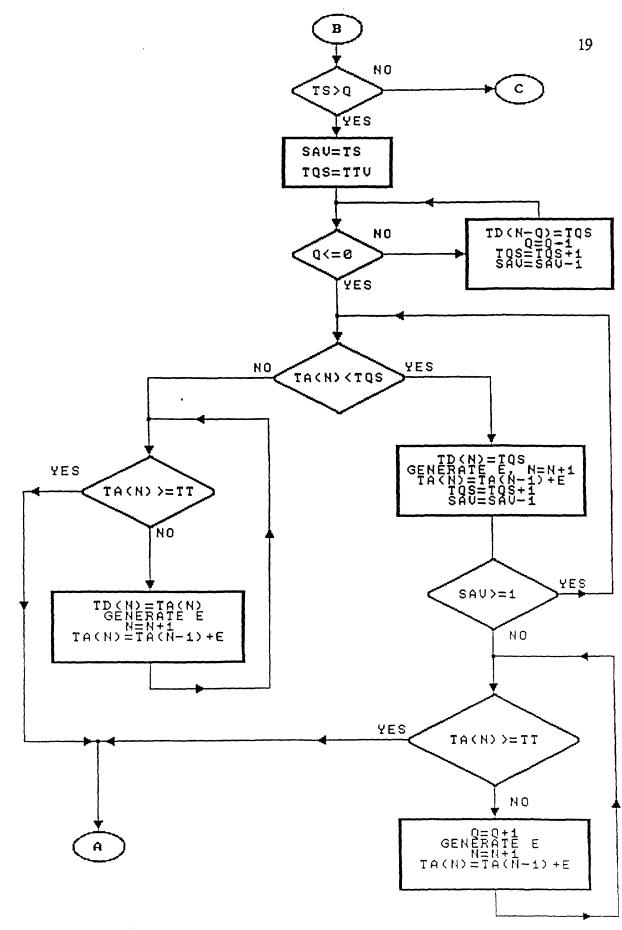


Figure 3.1(b) Voice/Data Integration Flowchart

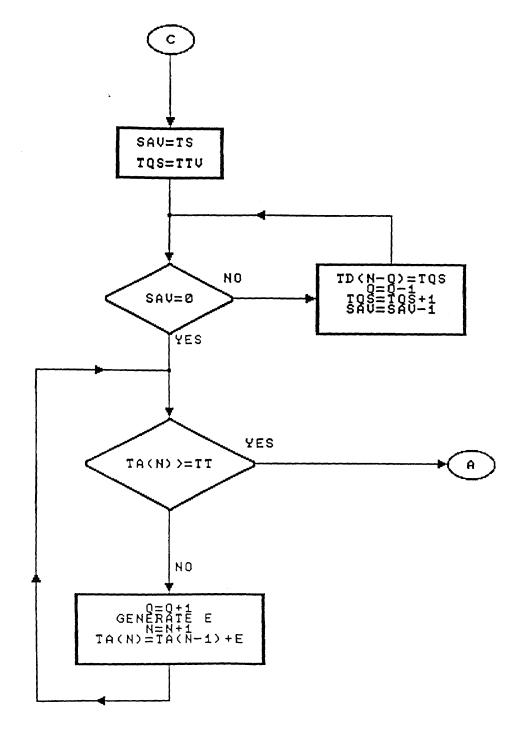


Figure 3.1(c) Voice/Data Integration Flowchart

duration. The arrival time of a packet is checked against talkspurt. If it has arrived during the talkspurt than it cannot be transmitted until the silence duration and therefore it is placed in the queue. All data packets arriving during talkspurt are placed in the queue and served in a First Come First Serve (FCFS) discipline during the a silence period. If a data packet arrives during silence period and if no data packet is waiting in queue, then that data packet is immediately served.

The primary statistical measurements, collected at the end of the simulation are average packet inter-arrival time, average talkspurt rate, probability of zero waiting time, and average data packet delay for different arrival rates.

3.4.3 PROGRAM

The FORTRAN source code is given in Appendix A.4.

CHAPTER 4

SIMULATION RESULTS

This chapter presents simulation results of various stochiastic processes including voice/data integration service. A comparison is made between simulation and theoretical values. Experiments for voice, silence duration and data packet arrival rate were run for 100,000 samples.

4.1 VOICE (TALKSPURT) SIMULATION RESULT

The voice (talkspurt) statistics, theoretical, and simulation results are shown in figure 4.1(a) and 4.1(b). Figure 4.1(a) presents the talkspurt length versus its corresponding probability of occurrence on a log scale. Figure 4.1(b) shows the same on a linear scale. The simulation curves are pretty much the same as the theoretical curves. A calculated average talkspurt length is 7.541 frames, where the simulated average talkspurt length is 7.547 frames.

4.2 VOICE INACTIVITY (SILENCE DURATION) SIMULATION RESULT

Silence duration statistics are presented in Figure 4.2(a) and 4.2(b), on a log scale and a linear scale respectively. Calculated and simulated values are nearly equal. This proves the validity of simulated silence duration statistics. As silence duration increases, the probability of silence duration (abscissa) exponentially decreases.

The value of simulated average silence duration is 5.953 frames while the calculated expected value is 6.659 frames.

One can see from figure 4.2(a) that the slope of the curve is different at lower values (< 8 frames) of silence duration and higher values (> 20 frames) of silence duration. Silence durations of less than 8 frames represent a silence periods between syllables and words while larger durations represent silence periods between phrases and sentences.

4.3 DATA PACKET ARRIVAL SIMULATION RESULTS

Results of the experiment with data packet arrival rates 4, 8, 12, and 16 packets/sec. are shown in figure 4.3 through figure 4.6. It is clearly seen from the figures that as arrival rate increases, the slope (negative) of probability versus interarrival time increases, which is expected. As arrival rate increase, packets are generated faster, therefore the time between two adjacent packets obviously decreases. For example, take a look at figure 4.6. We notice that at an arrival rate of 16 packets/sec, the probability of having the distance greater than 47 frames between the two adjacent data packets (inter-arrival time) is zero. Again, theoretical and simulation curves match very well.

Theoretical expected values of inter-arrival time are 11.11, 5.55, 3.709 and 2.77 frames for arrival rate of 4, 8, 12, 16 packets/second respectively. Corresponding values of simulated inter-arrival times are 11.12, 5.56, 3.707 and 2.78 frames.

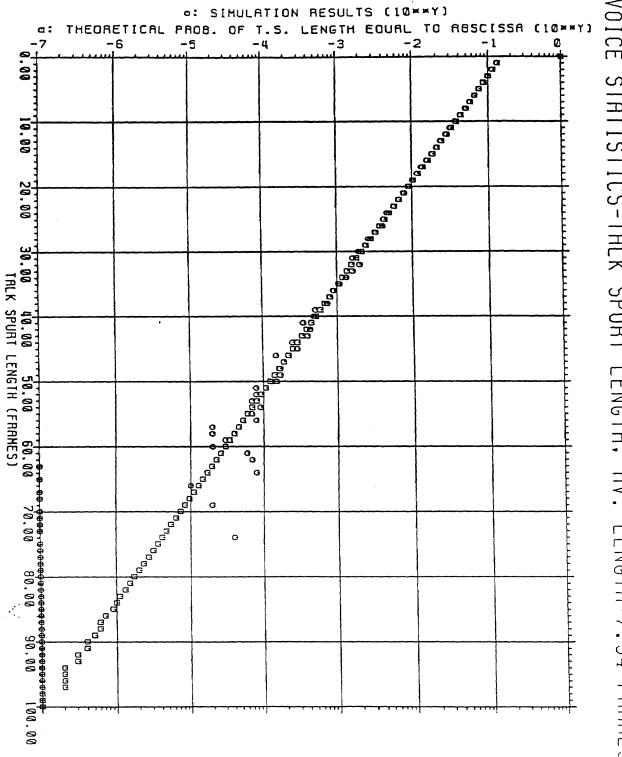


Figure 4.1(a) Voice Statistics (log scale)

VOICE STATISTICS-TALK SPURT LENGTH, AV. LENGTH=7.54 FRAMES

24

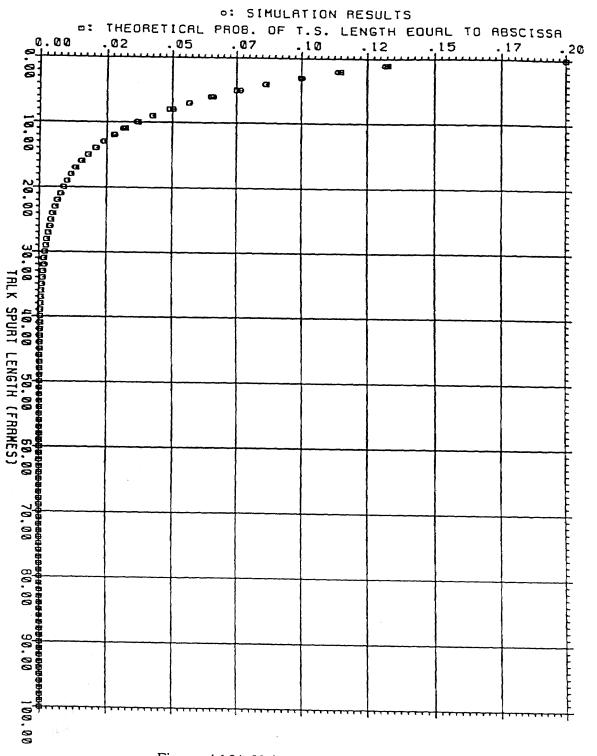


Figure 4.1(b) Voice Statistics (linear scale)

VOICE STATISTICS-TALK SPURT LENGTH, AV. LENGTH (I 7.54 FRAMES

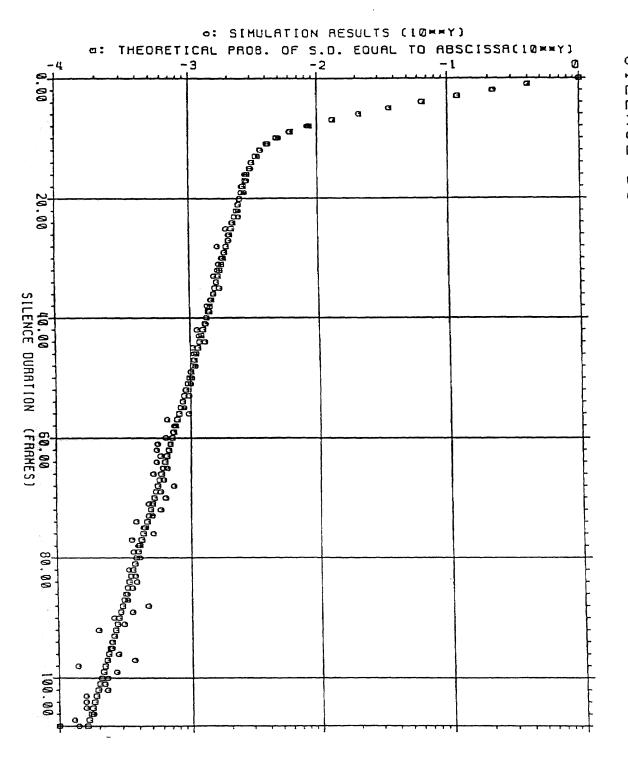


Figure 4.2(a) Silence Duration Statistics (log scale)

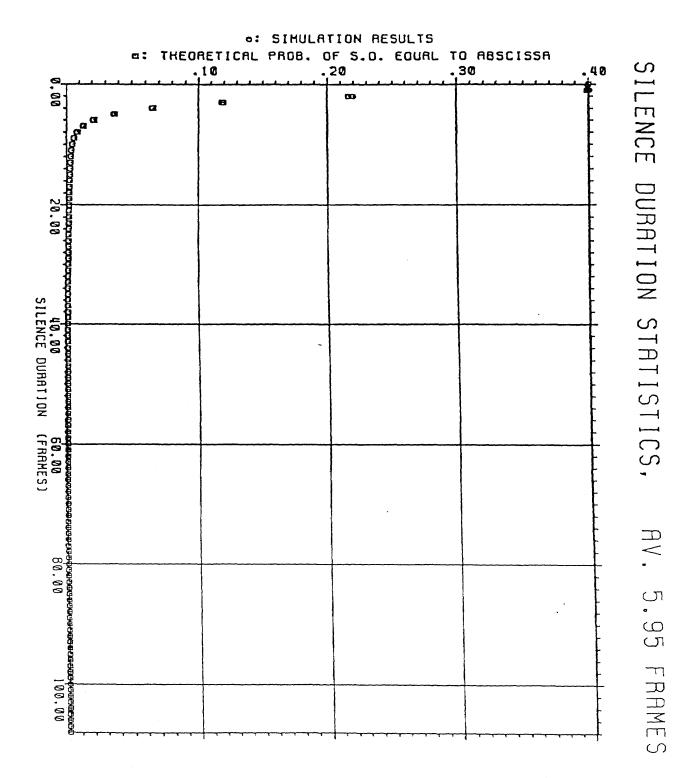


Figure 4.2(b) Silence Duration Statistics (linear scale)

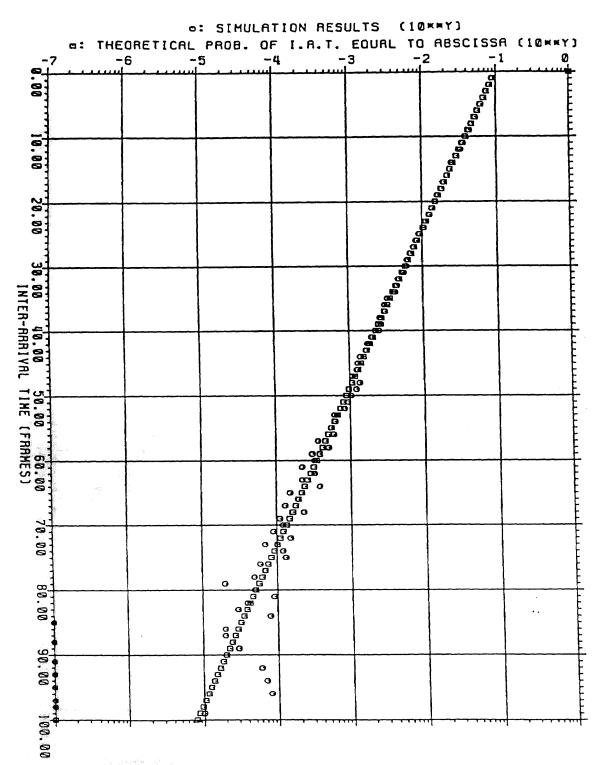


Figure 4.3 Data Packet Arrival Statistics at rate 4 packets/sec.

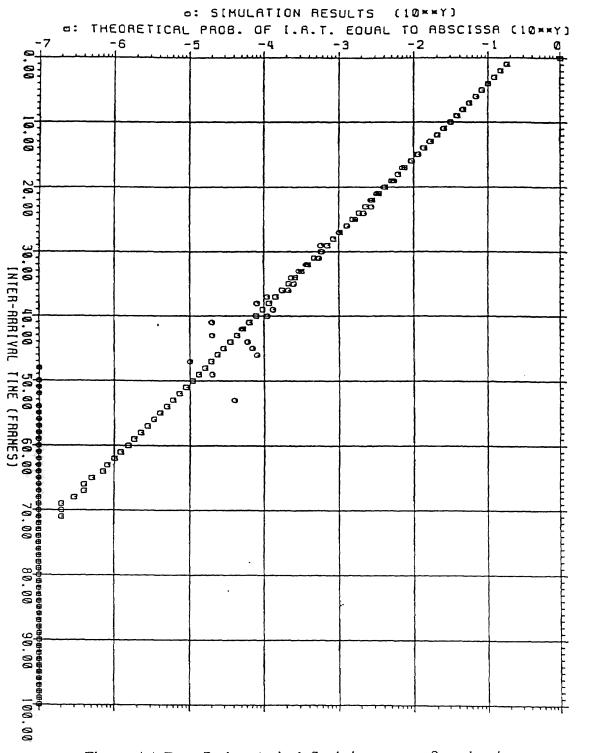


Figure 4.4 Data Packet Arrival Statistics at rate 8 packets/sec.

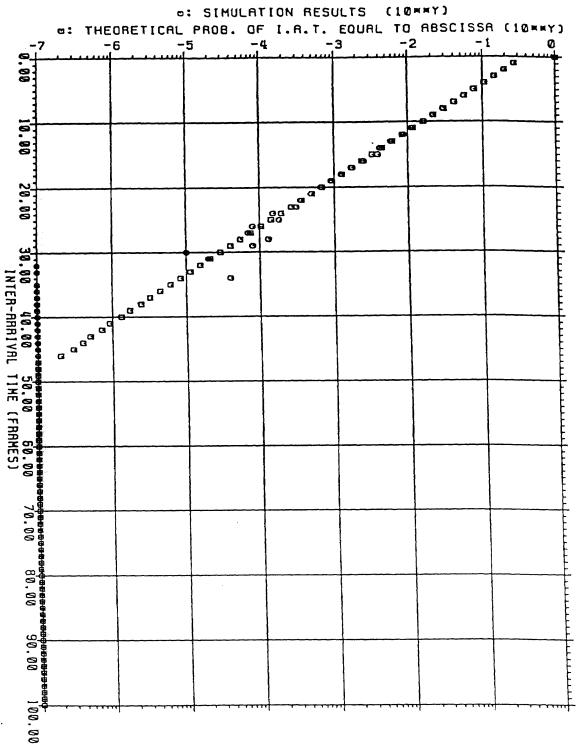


Figure 4.5 Data Packet Arrival Statistics at rate 12 packets/sec.

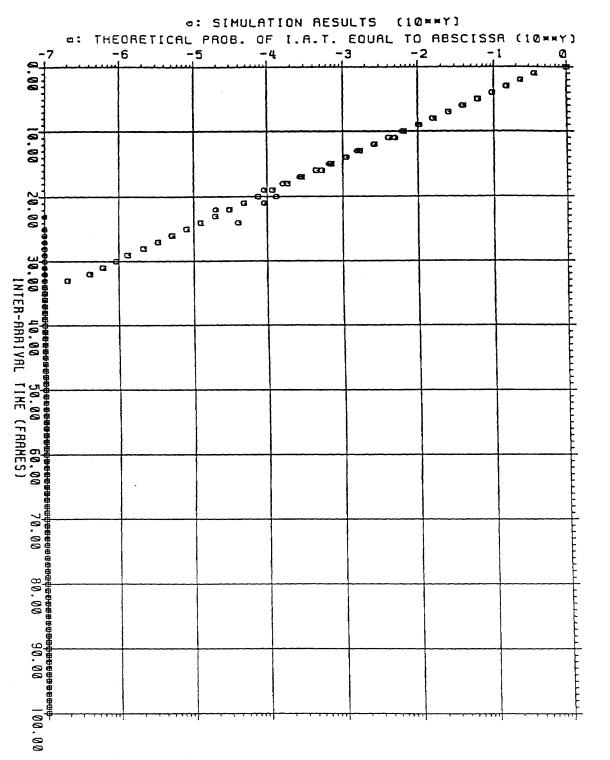


Figure 4.6 Data Packet Arrival Statistics at rate 16 packets/sec.

· . •.

. .

4.4 VOICE/DATA INTEGRATION SIMULATION RESULTS

The voice/data integration experiments were ran for 10,000 seconds (approximately 450,000 frames), and for various data arrival rate. Data arrival rates from 1 packet/second to 20 packets/second were used for simulation purpose. The following statistics were obtained for each case.

- 1. Total number of data packets generated.
- 2. Total number of data packets serviced.
- 3. Data packet arrival rate as per simulation.
- 4. Packet inter-arrival time (sec.).
- 5. Average talkspurt/silence rate (per sec.)
- 6. Probability of zero wait time.
- 7. Average data packet delay (frames).

8. Probability of wait time (0-200 frames).

The result of the sample run of this experiment is presented in appendix A.4. The seed value was 16807 for all experiments. Theoretical average talkspurt rate/silence rate is given by 1/(talkspurt + silence duration) = 3.13 /second. The simulated result is of 3.29 /second.

4.5 AVERAGE DATA PACKET DELAY VS. ARRIVAL RATE

The prime results of simulation are based on the above statistics. Simulation results of average data packet delay with different data packet arrival rates are shown in Table 4.1. Data packet delay is the time each packet spent waiting before

being serviced. This table shows the results for arrival rate from 1 packet/sec. to 20 packets/second. We can see, from the table as well as from figure 4.7 that average data packet delay increases with the arrival rate. Figure 4.7 shows that during the arrival rate 1 to 14 packets/second, average delay increases steadily. For an arrival rate of 15 packets/second and above, the average delay increases sharply, since it approaches data packet transmission capacity.

As the arrival rate increases, a greater amount of data packets come into the system. The service of these packets depends upon a silence duration statistic that has fixed capacity. The theoretical capacity for data packet transmission is: (Average silence duration)/[frame time*(Average talkspurt + Average silence duration)] = 6.659/[.0225*(6.659+7.541)] = 20.8 packets/second. The simulated data transmission capacity is: = 5.953/[.0225*(5.953+7.547)] = 19.6 packets/second. Therefore, all data packets cannot be served on their arrival, so they are placed in a queue. The packets in the queue wait until the silence duration occurs. As figure 4.7 states, when the arrival rate increases, the queue will increase and as well as the average data packet delay.

4.6 RATIO OF AVERAGE ARRIVAL RATE TO AVERAGE DELAY

Table 4.2 presents the data used to find the optimum arrival rate. The ratio of average arrival rate to average delay is considered as one of the performance measures. The performance curve is shown in figure 4.8. From this curve, optimum data packet arrival is 6 packets/second. There is not much performance degradation

TABLE : 4.1

VOICE/DATA INTEGRATION STATISTICS

SEED VALUE = 16807 AVERAGE TALK SPURT	TOTAL SIM. TIME (SEC) = 10000.00 RATE (PER SEC) = 3.29
AVERAGE DATA PACKET ARRIVAL RATE (PACKETS/SEC)	AVERAGE DATA PACKET DELAY (FRAMES)
1.000000	4.656875
2.000000	5.154598
3.000000	5.783758
4.000000	6.493093
5.00000	7.396003
6.000000	8.433848
7.000000	9.964211
8.000000	11.48726
9.000000	13.80577
10.00000	16.44546
11.00000	19.84325
12.00000	24.65047
13.00000	29.90484
14.00000	36.81536
15.00000	47.94444
16.00000	59.00859
17.00000	83.01635
18.00000	133.3712
19.00000	211.0172
20.00000	329.2377

na an an Anna a Anna an A

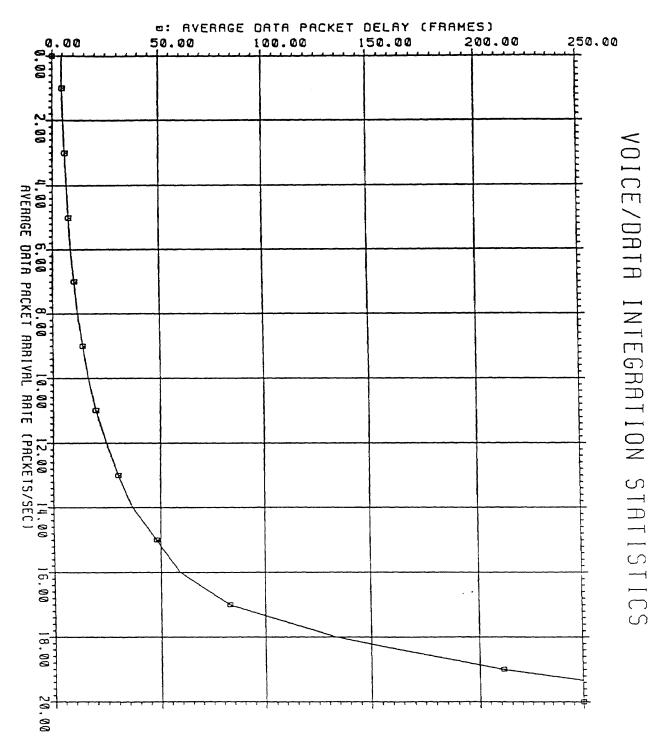


Figure 4.7 Voice/Data Integration Statistics

TABLE: 4.2

VOICE/DATA INTEGRATION STATISTICS

	RATE = 6.000000 R. RATE TO DELAY = 0.7114190
	RATIO OF AR.RATE
	0.2147363
1 2 3 4 5 6 7 8 9	0.3880031
3	0.5186939
4	0.6160392
5	0.6760408
6	0.7114190
7	0.7025142
8	0.6964237
	0.6519014
10	0.6080706
11	0.5543447 0.4868061
12 13	0.4347122
14	0.3802761
15	0.3128622
16	0.2711470
17	0.2047789
18	0.1349617
19	0.0900401
20	0.0607464

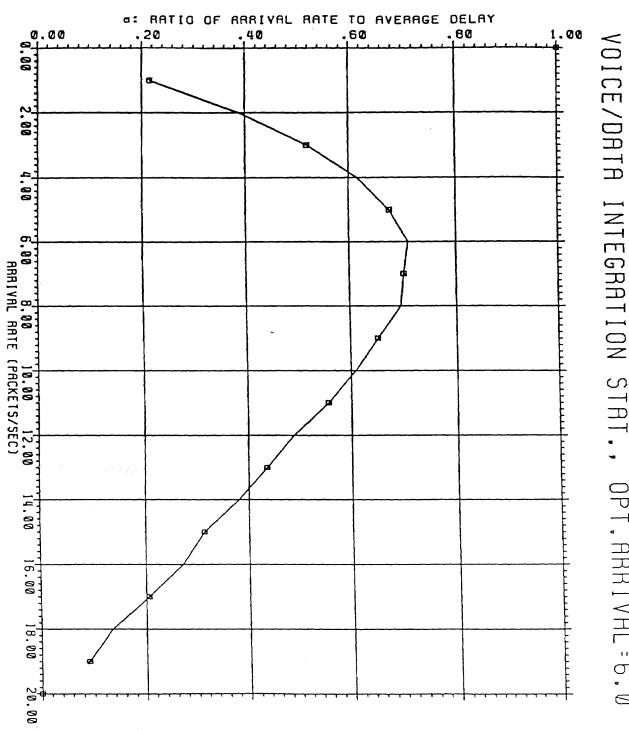


Figure 4.8 Optimum Arrival rate

INTEGRATION STAT., OPT.ARRIVAL:6.0

between 5 and 8 packets/second. The curve shows performance degradation with an arrival rate higher than 8 packets/second and less than 5 packets/second. Maximum system performance can be achieved with an arrival rate of 6 packets/second.

4.7 PROBABILITY OF WAIT TIME VS. ARRIVAL RATE

Results of probability delay statistics with four different arrival rates are presented in figures 4.9 to 4.12. Comparing these figures, it is obvious that the slope of the probability distribution flattens out as data packet arrival rate increases. That means the probability of any given delay increases with arrival rate. The probability of having a data packet delay of 50 frames has probability of 0.000035, 0.0013, 0.0035, and 0.0053 for arrival rates 4, 8, 12 and 16 packets/second respectively. Thus, average data packet delay also increases with arrival rate.

Average delay is equal to 6.44 frames at a rate of 4 packets/second. As arrival rate increases to 16 packets/second, then average delay jumps to 59 frames. Now, we can see that as arrival rate increases the probability of higher data packet delay also increases.

4.8 PROBABILITY OF ZERO WAIT TIME VS. ARRIVAL RATE

Table 4.3 shows voice and data integration statistics for zero waiting time. Zero waiting time means that an incoming packet is serviced immediately without waiting in the queue. This table and figure 4.13 shows the probability of zero waiting time decreases with increasing arrival rate. This means that the data packet will have to wait in queue for service. Chances of immediate service are reduced as arrival rate increases.

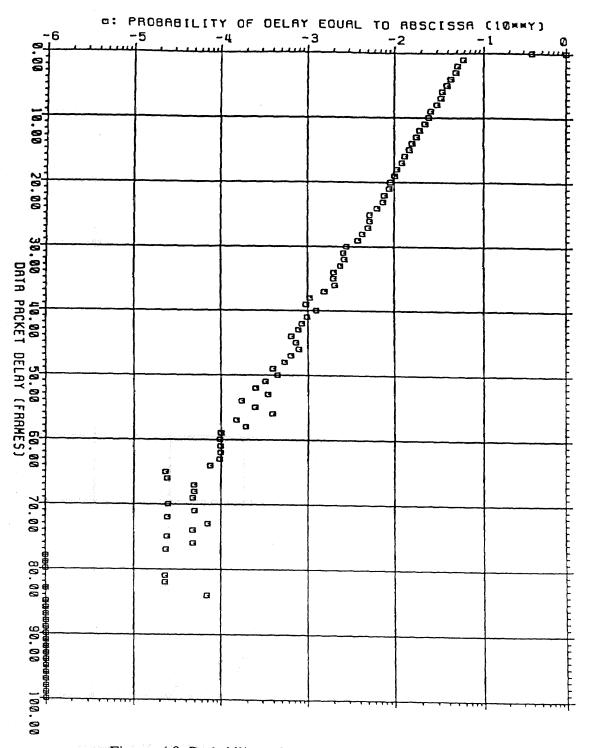
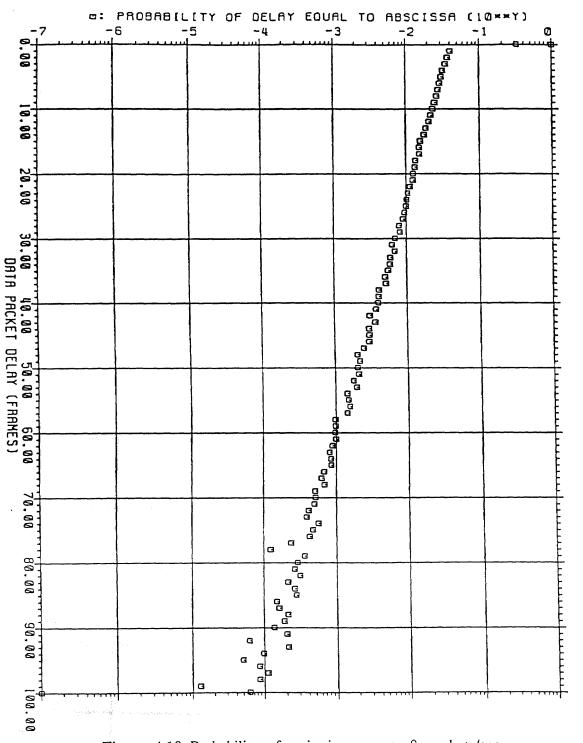
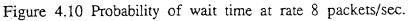


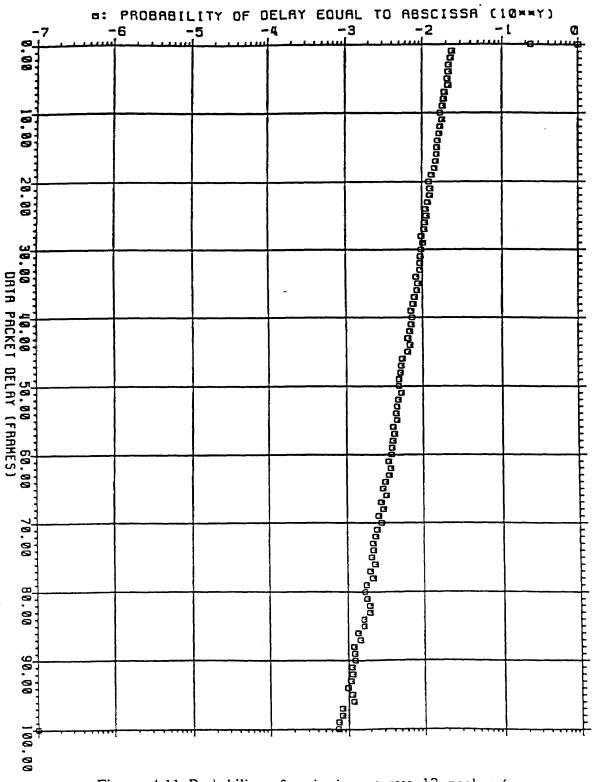
Figure 4.9 Probability of wait time at rate 4 packets/sec.

с. **Х**.





VOICE/DATA INTEGRATION STAT., AR.RATE:8, AV.DELAY:11.48



VOICE/DATA INTEGRATION STAT., AR. RATE:12, AV. DELAY:24.65

Figure 4.11 Probability of wait time at rate 12 packets/sec.

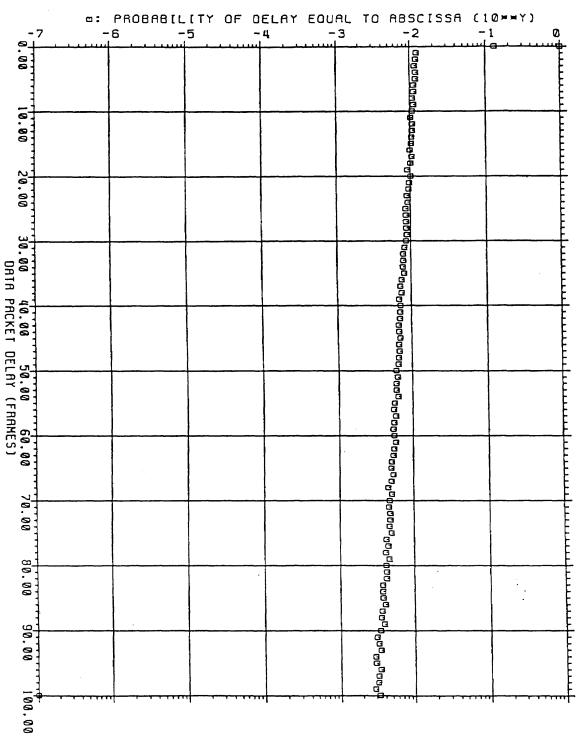


Figure 4.12 Probability of wait time at rate 16 packets/sec.

TABLE : 4.3

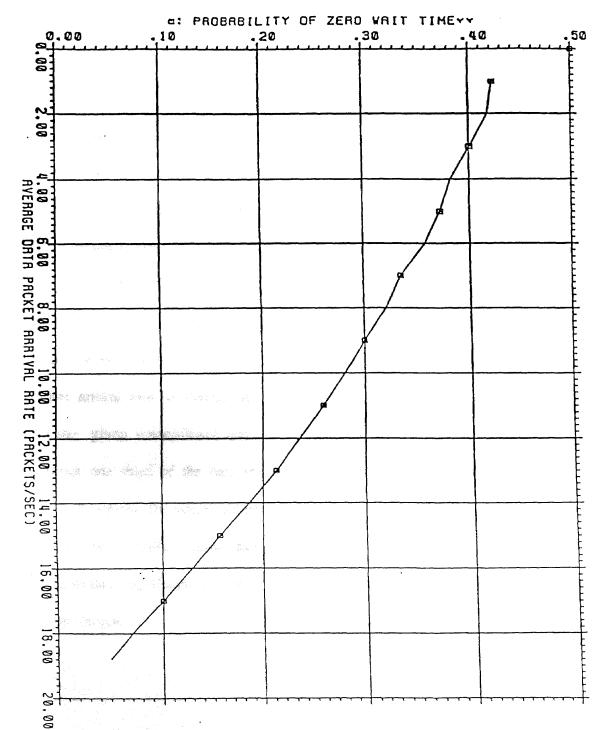
.

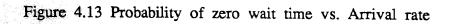
. ·

VOICE/DATA INTEGRATION STATISTICS

AVERAGE DATA PACKET ARRIVAL RATE (PACKETS/SEC)	PROBABILITY OF ZERO WAIT TIME
1.000000	0.4221061
2.000000	0.4176050
3.000000	0.4005996
4.000000	0.3824100
5.000000	0.3722043
6.00000	0.3580433
7.000000	0.3351898
8.00000	0.3207938
9.00000	0.2995426
10.00000	0.2797738
11.00000	0.2585041
12.00000	0.2348689
13.00000	0.2128907
14.00000	0.1861750
15.00000	0.1568836
16.00000	0.1295601
17.00000	0.1000240
18.00000	6.9931388E-02
19.00000	4.4454310E-02
20.0000	·· 2.6428798E-02







CHAPTER 5

CONCLUSION

A simulation study of several aspects of Voice/Data integration performance is presented. Voice (talkspurt) process is generated using geometric distribution. A silence duration process algorithm was developed by use of weighted geometric distribution. The data arrival process was also modelled using geometric distribution. The simulated outcome of all these models match closely with the theory.

The results of voice/data integration simulation shows that: (1.) Data packet delay increases (2.) Probability of zero wait time decreases and (3.) Probability of longer delay increases with increases in data packet arrival rate. The ratio of data packet arrival rate to average delay provides a performance measure of the system. For the given voice/silence statistics, the optimal arrival rate is 6 packets/sec. which is about one third of the data transmission capacity of the system.

A scheme for voice and data integration is provided which can be incorporated into a larger model. Simulation shows that the integrated voice/data system can accommodate significant growth in data traffic and still perform with acceptable data packet delays. APPENDIX A

SOURCE CODES

APPENDIX A.1

***** С С THIS PROGRAM GENERATE RANDOM NUMBER 00000 WITH "GEOMETRIC " PDF. "TALKSPURT" LENGTH IN "N" NUMBERS OF c c FRAMES. "N" IS GEOMETRICALLY DISTRIBUTED. С - URMI PATEL, FEB. 1991 č C** ******** С 000 THE PROGRAM PROMPTS FOR A SEED VALUE AND A NUMBER FOR TOTAL NUMBERS OF THE RANDOM NUMBERS TO BE GENERATED. IT ALSO ASKS FOR THE OUTPUT FILE NAME. c c IT GENERATES THE GEOMETRICALLY DISTRIBUTED RANDOM c c NUMBERS AND COMPUTES THEORETICAL AS WELL AS SIMULATED PROBABILITY DISTRIBUTION FUNCTIONS. С C* С IMPLICIT NONE INTEGER*4 IS, NT, N, IX(100), SUM RSUM, XT (100), X (100), RHO, R, RN, RG, XX REAL INTEGER K, NRG CHARACTER*12 FILENAME PRINT *, ' ENTER SEED VALUE AND TOTAL RANDOM NUMBERS' READ *, IS, NT READ *, IS, NT PRINT *, ' ENTER OUTPUT FILE NAME' READ (*, '(A)')FILENAME PRINT *, ' ENTER RHO' С READ *, RHO С С RHO=.8674 R=LOG (RHO) С С INITIALIZING THE VARIABLES С SUM=0 DO K=1,100 IX(K) = 0END DO С CALL SEED(IS) С DO N=1,NT CALL RANDOM (RN) ! UNIFORM RANDOM NUMBER (RN) GENERATION 50 IF (RN .LE. 0.0) GO TO 50 ! MAPPING "RN" TO GEOMETRIC DISTRIBUTED RG = (LOG(RN)/R) + .5! "NRG". NRG =NINT (RG) SUM=SUM+NRG C COMPUTING PROBABILITY DISTRIBUTION FUNCTION С С DO K=1,100 IF (NRG .EQ. K) IX(K) = IX(K) + 1END DO C END DO RSUM=SUM PRINT *, ' AVERAGE TALK SPURT LENGTH = ', RSUM/NT С OPEN (UNIT=3, FILE=FILENAME, STATUS=' UNKNOWN') · • •

WRITE(3,*)' VOICE ACTIVITY (TALK SPURT) STATISTICS'
WRITE(3,*)''
WRITE(3,*)' SEED VALUE = ', IS,' TOTAL NUMBERS = ', NT
WRITE(3,*)' AVERAGE TALK SPURT LENGTH (FRAMES) = ', RSUM/NT
WRITE(3,*)''
DO K=1,100
XT(K) = (1-RHO)*RHO**(K-1) ! THEORETICAL DISTRIBUTION
XX=IX(K)
X(K)=XX/NT
PRINT *, K, XT(K), X(K)
WRITE(3,200)K, XT(K), X(K)
FORMAT(2X, I5, 2X, 2F12.7)
END DO
CLOSE (UNIT=3)
STOP
END

200

ť

•

APPENDIX A.2

C C THIS PROGRAM GENERATE RANDOM NUMBER WITH "Weighted GEOMETRIC " PDF. "SILENCE DURATION" IS "N" FRAME LONG. THE DISTRIBUTION OF THE "N" IS WEIGHTED GEOMETRIC. - URMI PATEL, FEB. 1991 Č* C ************************* 00000 THIS PROGRAM PROMPTS FOR A SEED VALUE AND A NUMBER FOR TOTAL RANDOM NUMBERS TO BE GENERATED. IT ALSO ASKS FOR A OUTPUT FILE. IT GENERATES WEIGHTED GEOMETRICALLY DISTRIBUTED C C NUMBERS AND COMPUTES THEORETICAL AS WELL AS SIMULATED PROBABILITY DISTRIBUTION. С C* С IMPLICIT NONE INTEGER*4 IS, NT, N, IX(108), SUM INTEGER K, J, TS REAL C1, C2, R1, R2, RN, GJ1, GJ RSUM, XT (108), X (108), XX REAL. CHARACTER*12 FILENAME PRINT *,' ENTER SEED VALUE AND TOTAL RANDOM NUMBERS' READ *, IS, NT PRINT *, ' ENTER OUTPUT FILE NAME' READ (*, '(A)') FILENAME С С C1=.8746 C1=.8531 R1=.5372 C2=.1469 R2=.9695 С č INITIALIZING THE VARIABLES SUM=0 DO K=1,108 IX(K)=0END DO С CALL SEED(IS) С DO N=1,NT 90 CALL RANDOM (RN) ! UNIFORM RANDOM NUMBER (RN) GENERATED IF (RN .LE. 0.0) GO TO 90 С С MAPPIN OF "RN" IN TO WEIGHTED GEOMETRICALLY DISTRIBUTED RANDOM NUMBER. (NEXT 8 LINES) С J=1 IF (J .GE. 109) GO TO 90 110 GJ1=C1+C2-(C1*R1**(J-1)+C2*R2**(J-1)) GJ = C1 + C2 - (C1 + R1 + J + C2 + R2 + J)IF ((RN .GE. GJ1) .AND. (RN .LT. GJ))GO TO 121 J=J+1 GO TO 110 121 TS=J • .

с	
-	SUM=SUM+TS
C	COMPUTING SIMULATED DISTRIBUTION FUNCTION DO K=1,108 IF (TS .EQ. K) IX (K) =IX (K) +1
	END DO
с	END DO RSUM=SUM PRINT *,' AVERAGE SILENCE DURATION = ',RSUM/NT
с	
	OPEN (UNIT=3, FILE=FILENAME, STATUS='UNKNOWN') WRITE (3,*)' SILENCE DURATION STATISTICS' WRITE (3,*)' ' WRITE (3,*)' SEED VALUE = ',IS,' TOTAL NUMBERS = ',NT WRITE (3,*)' AVERAGE SILENCE DURATION (FRAMES) =',RSUM/NT WRITE (3,*)' ' DO K=1,108 XT (K)=C1*(1-R1)*R1**(K-1)+C2*(1-R2)*R2**(K-1) ! THEORETICAL XX=IX (K) X (K)=XX/NT DDINE * K VE(K) X(K)
С	PRINT *,K,XT(K),X(K) WRITE(3,200)K,XT(K),X(K)
200	

.•

APPENDIX A.3

C С THIS PROGRAM GENERATE RANDOM NUMBER С č WITH "GEOMETRIC " PDF FOR DATA PACKET C C INTER-ARRIVAL TIME. ("N" FRAME LONG) C C "N" IS GEOMETRICALLY DISTRIBUTED. С - URMI PATEL, FEB. 1991 С С С THE PROGRAM PROMPTS FOR A SEED VALUE AND A Ċ NUMBER FOR TOTAL NUMBERS OF THE RANDOM NUMBERS TO BE С GENERATED. IT ALSO ASKS FOR AN AVERAGE DATA PACKET C C ARRIVAL RATE (LAMBDA) AND A OUTPUT FILE NAME. EACH PACKET IS ONE FRAME (22.5 MILLI-SECOND) LONG. AN INTER-ARRIVAL TIME IS INTEGRAL MULTIPLE OF FRAME 0000000 TIME. IT GENERATES THE GEOMETRICALLY DISTRIBUTED RANDOM NUMBERS AND COMPUTES THEORETICAL AS WELL AS SIMULATED PROBABILITY DISTRIBUTION FUNCTIONS. С ****** C* С IMPLICIT NONE INTEGER*4 IS, NT, N, IX(100), SUM REAL RSUM, XT (100), X (100), LAMBDA, RN, RG, RHO, R, XX INTEGER K, NRG CHARACTER*12 FILENAME PRINT *,' ENTER SEED VALUE, TOTAL RANDOM NUMBERS AND' PRINT *,' AVERAGE DATA PACKET ARRIVAL RATE (LAMBDA).' READ *, IS, NT, LAMBDA PRINT *,' ENTER OUTPUT FILE NAME' READ (*,'(A)')FILENAME С С RHO=1.-.0225*LAMBDA R=LOG (RHO) C C INITIALIZING VARIABLES С . SUM=0 DO K=1,100 IX(K) = 0END DO С CALL SEED(IS) С DO N=1,NT 50 CALL RANDOM (RN) ! UNIFORM RANDOM NUMBER (RN) GENERATION IF (RN .LE. 0.0) GO TO 50 ! MAPPING "RN" TO GEOMETRIC DISTRIBUTED RG=(LOG(RN)/R)+.5NRG =NINT (RG) ! "NRG". SUM=SUM+NRG С COMPUTING PROBABILITY DISTRIBUTION FUNCTION ¢ С DO K=1,100 IF (NRG .EQ. K) IX (K) = IX (K) +1 END DO С

```
END DO

RSUM=SUM

PRINT *,' AVERAGE INTER-ARRIVAL TIME (FRAMES) = ',RSUM/NT

OPEN (UNIT=3,FILE=FILENAME, STATUS='UNKNOWN')

WRITE (3,*)' DATA PACKET ARRIVAL STATISTICS'

WRITE (3,*)' WRITE (3,*)' SEED VALUE = ',IS,' TOTAL NUMBERS = ',NT

WRITE (3,*)' AVERAGE DATA PACKET ARRIVAL RATE =',LAMBDA

WRITE (3,*)' AVERAGE INTER-ARRIVAL TIME (FRAMES) = ',RSUM/NT

WRITE (3,*)' '

DO K=1,100

XT (K) = (1-RHO)*RHO**(K-1) ! THEORETICAL DISTRIBUTION

XX=IX (K)

X (K) = XX/NT

PRINT *,K,XT (K),X (K)

FORMAT (2X,I5,2X,2F12.7)

END DO

CLOSE (UNIT=3)

STOP

END
```

200

С

÷

С С THIS PROGRAM SIMULATES VOICE/DATA INTEGRATION С IN DATA COMMUNICATION SYSTEM. 00000 THE DATA PACKETS ARE TRANSMITED DURING INACTIVITY OF VOICE CHANNEL (i.e. DURING SILENCE). URMI PATEL c c FEB. 1991. NEW JERSEY INSTITUTE OF TECHNOLOGY С С A DATA PACKETS ARE ASSUMED TO BE ONE FRAME LENTH LONG. Ċ AN INTER ARRIVAL TIME OF DATA PACKET IS ASSUMED TO BE С GEOMETRICALLY DISTRIBUTED. C C A VOICE IS PACKETIZED. ITS LENTH IS EXPRESSED IN AN INTEGRAL NUMBER OF FRAMES. A FRAME IS 22.5 MILLI-SEC LONG. A VOICE TALK-SPURT LENTH (IN TERMS OF FRAMES) 0000 IS GEOMETRICALLY DISTRIBUTED. С A SILENCE DURATION CAN ALSO BE EXPRESSED IN TERMS OF FRAMES. THE DURATION (IN TERMS OF FRAMES) FOLLOWS Ċ С WEIGHTED GEOMETRICAL DISTRIBUTION. С С С THIS PROGRAM PROMPTS FOR SEED VALUE, MAXIMUM SIMULATION TIME AND PACKET ARRIVAL RATE (LAMBDA). С IT ALSO ASKS AN OUTPUT FILE NAME WHERE PACKET DELAY С DISTRIBUTION AND OTHER STATISTICS BE STORED. С С С IMPLICIT NONE INTEGER*4 TA (0:15000), TD (0:15000) INTEGER*4 IS, NMAX, TT, TTV, TQS, N, SUMD, NAR, NSERV INTEGER*4 WNAR, WNSERV INTEGER E, TV, TS, SAV, L, I, J, K, M, NREP INTEGER Q, V, IX (0:200), DLY (0:15000) LAMBDA, TMAX, RSUMD, TN, VV, X (0:200) REAL APAR, APIAT, ATSR, ADPD, XX REAL REAL WAPAR, WAPIAT, WATSR, WADPD, WX (0:200) COMMON LAMBDA, E, TV, TS, IS CHARACTER*15 FILENAME С С PROMPTS FOR INPUTS č PRINT *, ' ENTER SEED VALUE, MAX. SIMULATION TIME AND' PRINT *, ' PACKET ARRIVAL RATE (LAMBDA) ' READ *, IS, TMAX, LAMBDA PRINT *,' ENTER OUTPUT FILE NAME' READ(*,'(A)')FILENAME С ! Used in micro-soft FORTRAN CALL SEED (IS) С NMAX=10000 NREP=NINT((TMAX/225.)+.5) ! .0225*NMAX = 225 С С INITIALIZES LONG LOOP С WNAR=0 WNSERV=0

WAPIAT=0. WADPD=0. DO K=0,200 X(K) = 0. END DO С DO M=1,NREP ! LONG LOOP STARTS.... 000 INITIALIZES VARIABLES FOR MAIN PROGRAM N=0 Q=0 V=0 TT=0 TTV=0 TQS=0 TA(N) = 0TD(N)=0CALL ARRIVAL (LAMBDA, E) N=N+1 TA(N) = TA(N-1) + EС С С С MAIN PROGRAM STARTS..... IF (TA(N) .GE. NMAX) GO TO 800 CALL VOICE(TV) CALL SILENCE(TS) 20 TTV=TT+TV TT=TT+TV+TS V=V+1 C 50 IF (TA (N) .GE. TTV) GO TO 100 Q=Q+1 CALL ARRIVAL (LAMBDA, E) N=N+1 TA(N) = TA(N-1) + EGO TO 50 С 100 IF (TS.GT.Q) GO TO 500 SAV=TS TQS=TTV С 150 IF (SAV.LE.0) GO TO 200 TD(N-Q) = TQSQ=Q-1 TQS=TQS+1 SAV=SAV-1 GO TO 150 С 200 IF (TA (N).GE.TT) GO TO 20 Q=Q+1 CALL ARRIVAL (LAMBDA, E) N=N+1TA(N) = TA(N-1) + EGO TO 200 ċ 500 SAV=TS TQS=TTV IF (Q.LE.0) GO TO 600 550 TD(N-Q) = TQSQ=Q-1 TQS=TQS+1 SAV=SAV-1 GO TO 550

WAPAR=0.

С

"我们的过去了我"

600 650	IF (TA (N) .LT. TQS) GO TO 700 IF (TA (N) .GE. TT) GO TO 20 TD (N) =TA (N) CALL ARRIVAL (LAMBDA, E) N=N+1 TA (N) =TA (N-1) +E GO TO 650
C 700	TD (N) =TQS CALL ARRIVAL (LAMBDA, E) N=N+1 TA (N) =TA (N-1) +E
	TQS=TQS+1 SAV=SAV-1
c	IF (SAV .GT. 0) GO TO 600
C 750	IF (TA (N) .GE. TT) GO TO 20 Q=Q+1 CALL ARRIVAL (LAMBDA, E) N=N+1 TA (N) =TA (N-1) +E GO TO 750
C 800	CONTINUE
	MAIN PROGRAM ENDS.
C C	DELAY STATISTICS CALCULATION STARTS
•	DO L=0,200 IX(L)=0 END DO
С	SUMD=0 NSERV=N-1-0
с	NSERV=N-1-0 Do I=1,NSERV
С	DLY (I) =TD (I) -TA (I) PRINT *, TD (I), TA (I), DLY (I) SUMD=SUMD+DLY (I)
C C	DELAY DISTRIBUTION CALCULATION J=0
1100	IF(J .GT. 200)GO TO 1201 IF(DLY(I) .EQ. J)GO TO 1200 J=J+1
1201	GO TO 1100 · IX(J)=IX(J)+1 CONTINUE
C .	END DO
C C C	DELAY STATISTICS CALCULATION ENDS.
C	RSUMD=SUMD NAR=N-1 TN=NAR VV=V APAR=TN/(0.0225*NMAX) APIAT=1./APAR ATSR=VV/(0.0225*NMAX) ADPD=RSUMD/NSERV
	DO K=0,200 XX=IX(K)

56

•

. .

X(K)=XX/NSERV END DO С ĉ ACCUMULATES RESULT OF REPEATED CALCULATIONS WNAR=WNAR+NAR WNSERV=WNSERV+NSERV WAPAR=WAPAR+APAR WAPIAT=WAPIAT+APIAT WATSR=WATSR+ATSR WADPD=WADPD+ADPD DO K=0,200 WX(K) = WX(K) + X(K)END DO С END DO С С LONG LOOP ENDS. С С NORMALIZES THE RESULTS С WAPAR=WAPAR/NREP WAPIAT=WAPIAT/NREP WATSR=WATSR/NREP WADPD=WADPD/NREP DO K=0,200 WX (K) =WX (K) /NREP END DO С С DATA PROCESSING COMPLETE. С PRINT *, ' TOTAL NUMBER OF DATA PACKETS GENERATED =', WNAR PRINT *,' TOTAL NUMBER OF DATA PACKETS SERVICED =', WNSERV PRINT *,' AVERAGE PACKET ARRIVAL RATE (PER SEC) = ', WAPAR PRINT *,' AVERAGE PACKET INTER-ARRIVAL TIME (SEC) = ', WAPIAT PRINT *, AVERAGE TALK SPURT RATE (PER SEC) = ', WATSR PRINT *, ' PROBABILITY OF ZERO WAITING TIME =', WX(0) PRINT *, ' AVERAGE DATA PACKET DELAY (FRAMES) =', WADPD С OPEN (UNIT=3, FILE=FILENAME, STATUS='UNKNOWN') VOICE/DATA INTEGRATION STATISTICS' WRITE (3, *) WRITE (3, *) ' WRITE (3,*)' ' WRITE (3, *)' SEED VALUE =', IS,' TOTAL SIM. TIME (SEC) =', TMAX WRITE (3, *)' AVERAGE DATA PACKET ARRIVAL RATE =', LAMEDA WRITE (3,*)' WRITE (3, *)' TOTAL NUMBER OF DATA PACKETS GENERATED =', WNAR WRITE (3, *)' TOTAL NUMBER OF DATA PACKETS GENERATED =', WNSERV WRITE (3, *)' AVERAGE PACKET ARRIVAL RATE (PER SEC) =', WAPAR WRITE (3, *)' AVERAGE PACKET INTER-ARRIVAL TIME (SEC) =', WAPIAT WRITE (3, *)' AVERAGE TALK SPURT RATE (PER SEC) =', WATSR WRITE (3, *)' PROBABILITY OF ZERO WAITING TIME =', WX (0) WRITE (3, *)' AVERAGE DATA PACKET DELAY (FRAMES) =', WADPD WRITE (3, *)' DO K=0,200 PRINT *, K, WX (K) C WRITE (3, 2000) K, WX (K) 2000 FORMAT (2X, 15, 2X, F12.7) END DO CLOSE (UNIT=3) STOP END

· . .

C******** 00000000 THIS SUBROUTINE PROGRAM GENERATE RANDOM NUMBER WITH "GEOMETRIC " PDF. "ARRIVAL" C* ******************************* С SUBROUTINE ARRIVAL (LAMBDA, E) INTEGER E REAL LAMBDA С Ċ C FRAME TIME IS 0.0225 SEC RHO=1.-.0225*LAMBDA С 950 ! Used in Microsoft FORTRAN CALL RANDOM (RN) RN=RAN (IS) IF (RN .LE. 0.0) GO TO 950 RG= (LOG (RN) /LOG (RHO))+.5 C950 ! Used in VAX FORTRAN E=NINT (RG) RETURN END

000000 THIS SUBROUTINE PROGRAM GENERATE RANDOM NUMBER WITH "GEOMETRIC " PDF. TALK SPURT LENGTH (IN NUMBERS OF FRAMES) Ĉ C** ************* SUBROUTINE VOICE (TV) INTEGER TV С RHO=.8674 с 960 ! Used in Microsoft FORTRAN ! Used in VAX FORTRAN CALL RANDOM (RN) RN=RAN(IS) C960 IF (RN .LE. 0.0)GO TO 960 RG=(LOG(RN)/LOG(RHO))+.5 TV=NINT(RG) RETURN END · .

0	*************	
С		
с с	THIS SUBROUTINE PROGRAM GENERATE RANDOM	
č	NUMBER WITH "WEIGHTED GEOMETRIC " PDF.	
č		
000	SILENCE DURATION (IN NUMBERS OF FRAMES)	
C		
C*****	************	
	SUBROUTINE SILENCE (TS) INTEGER TS	
С		
ċ	C1=.8746	
•	C1=.8531	
	C2 = .1469	
	RHO1=.5372	
	RH02= . 9695	
с		
70	CALL RANDOM (RN) ! Used in Microsoft FORTR	AN
C70	RN=RAN(IS) ! Used in VAX FORTRAN	
0.0	IF (RN .LE. 0.0) GO TO 70	
	J=1	
100	IF(J.GT. 108)GO TO 70	•
100	GJ1=C1+C2-(C1*RHO1**(J-1)+C2*RHO2**(J-1))	
	GJ=C1+C2-(C1*RH01**J+C2*RH02**J)	
	GO = CI + C2 = (CI = Riloi = O + C2 = Rilo2 = O) IF ((RN .GE.'GJ1) .AND. (RN .LT. GJ))GO TO 101	
	J=J+1	
	GO TO 100	
1 0 1		
101	TS=J	
	RETURN	
	END	

. ·

SEED VALUE = 16807 TOTAL SIM. TIME (SEC) = 10000.00 AVERAGE DATA PACKET ARRIVAL RATE = 8.000000 TOTAL NUMBER OF DATA PACKETS GENERATED = 81070 TOTAL NUMBER OF DATA PACKETS SERVICED = 81010

TOTAL NUMBER OF DATA PACKETS TOTAL NUMBER OF DATA PACKETS AVERAGE PACKET ARRIVAL RATE AVERAGE PACKET INTER-ARRIVAL AVERAGE TALK SPURT RATE (PER PROBABILITY OF ZERO WAITING AVERAGE DATA PACKET DELAY (F	SERVICED (PER SEC) = TIME (SEC SEC) = 3 TIME = 0.	= 81010 8.006913
$\begin{array}{cccccccccccccccccccccccccccccccccccc$	52 53 4 55 56 7 58 9 0 12 3 4 56 7 7 7 7 7 7 7 7 7 7 7 7 7 7 7 7 7 7	0.0017867 0.0019525 0.0014397 0.0014336 0.0015548 0.0015548 0.0009768 0.0009768 0.0009827 0.0009863 0.0008023 0.0008413 0.0006688 0.0006196 0.0006701 0.0005057 0.0005074 0.0005057 0.0005074 0.0004099 0.0003858 0.000527 0.0004682 0.0004682 0.0004682 0.0002331 0.0002331 0.0002331 0.0002331 0.0002580 0.0002580 0.0002580 0.0002574 0.0002580 0.0002574 0.0002574 0.0002574 0.000290 0.0002574 0.000291 0.000291 0.0001583 0.000202 0.000202 0.0001583 0.000202 0.000202 0.000202 0.000202 0.000202 0.000202 0.000202 0.000202 0.000202 0.00002707 0.0000497 0.0000497 0.0000497 0.0000497 0.0000497 0.0000202 0.0000241 0.0000241 0.0000241 0.0000500 0.0000737 0.0000489 0.0000500 0.0000737 0.0000489 0.0000241 0.0000000 0.0000000 0.0000000

1110122345- 11102122345- 111111111111111111111111111111111111	0.0000244 0.0000124 0.0000124 0.0000122 0.0000122 0.0000122 0.0000122 0.0000122 0.0000124 0.0000124 0.0000124 0.0000124 0.00000124 0.0000000 0.0000000 0.0000000 0.0000000
178 179 180 181 182 183	$\begin{array}{c} 0.0000000\\ 0.0000000\\ 0.0000000\\ 0.0000000\\ 0.0000000\\ 0.0000000\\ 0.0000000\end{array}$

2.04	
184	0.000000
185	0.0000000
186	0.0000000
187	0.0000000
188	0.0000000
189	0.0000000
190	0.0000000
191	0.0000000
192	0.000000
193	0.0000000
194	0.000000
195	0.000000
196	0.0000000
197	0.0000000
198	0.0000000
199	0.0000000
200	0.0000000

APPENDIX B

DERIVATION OF EXPECTED VALUE

.

Expected Value of Silence Duration:

.

. .

The probability of silence duration is given by weighted geometric function as below:

$$f_{I}(k) = C_{1} (1-r_{1}) r_{1}^{k-1} + C_{2} (1-r_{2}) r_{2}^{k-1}$$

$$E(X) = \sum_{k=1}^{\infty} \{k[C_{1} (1-r_{1}) r_{1}^{k-1} + C_{2} (1-r_{2}) r_{2}^{k-1}]\}$$

$$= C_{1} \sum_{k=1}^{\infty} k (1-r_{1}) r_{1}^{k-1} + C_{2} \sum_{k=1}^{\infty} k (1-r_{2}) r_{2}^{k-1}$$

$$= C_{1} (1-r_{1}) \sum_{k=1}^{\infty} kr_{1}^{k-1} + C_{2} (1-r_{2}) \sum_{k=1}^{\infty} k r_{2}^{k-1}$$

Let
$$S = \sum_{k=1}^{\infty} k r_1^{k-1} = 1 + 2r^1 + 3r^2 + 4r^3 + \dots (2)$$

 $r_1 S = \sum_{k=1}^{\infty} k r_1^k = r + 2r^2 + 3r^3 + 4r^4 + \dots (3)$

Subtracting equation (3) from equation (2) we get:

$$S - r_1 S = 1 + r^1 + r_1^2 + r_1^3 + r_1^4....(4)$$

$$S (1 - r_1) = 1 / (1 - r_1)$$

$$S = 1 / (1 - r_1)^2$$

$$E(X) = C_1 - \frac{(1-r_1)}{(1-r_1)^2} + C_2 \frac{(1-r_2)}{(1-r_2)^2}$$

$$=\frac{C_1}{(1-r_1)} + \frac{C_2}{(1-r_2)}$$
.....(5)

.

E(X) is a theoretical expected value of silence duration. For $C_1 = .8531$, $C_2 = .1469$, $r_1 = .5372$ and $r_2 = .9695$,

E(X) = 6.659 frames.

Expected Value of Talkspurt:

If $C_1 = 1$, $C_2 = 0$ and $r_1 = r$ than, equation (5) becomes:

E(X) = 1/(1 - r)

.

for r = .8675,

E(X) = 7.541 frames.

SELECTED BIBLIOGRAPHY

- [1] Brady, Paul T., "A Technique for Investigating On-Off Patterns of Speech," <u>The</u> <u>Bell System Technical Journal</u>, vol. 44, January, 1965, pp. 1-22.
- [2] Byron, J. T. Morgan, <u>Elements of Simulation</u>. New York: Chapman and Hall, 1984, pp. 91-95.
- [3] Cox, D. R. and Smith, Walter, L., Queues. London: Chapman and Hall, 1979.
- [4] Gruber, John G., "A Comparison of Measured and Calculated Speech Temporal Parameters Relevant to Speech Activity Detection," <u>IEEE Transaction</u> on Communications, Vol. COM-30, April, 1982, pp. 728-738.
- [5] Hastings, N. A. J. and Peacock, J. B., <u>Statistical Distributions</u>. New York: Halsted Press, 1975, pp. 74-77.
- [6] Schwartz, Mischa, <u>Telecommunication Networks: Protocols</u>, <u>Modelling and</u> <u>Analysis</u>. New York: Eddison-Wesley Publishing Company, 1987.
- Sherman, D. N., "Storage and Delay Estimates for Asynchronous Multiplexing of Data in Speech," <u>IEEE Transaction on Communication Technology</u>, August, 1971, pp. 551-555.
- [8] Stallings, William, <u>Data and Computer Communications</u>. New York: Macmillan Publishing Company, 1988, pp. 697–705.
- [9] Stern, H. P., "Design and Performance of an Integrated Voice/Data Mobile Radio System," <u>IEEE Globe COM-89</u>," Dallas, Texas, March, 1989, pp. 433– 437.